Voice Mail Administrator Features
If your voice mail system has Automatic Speech Recognition (ASR) capability and it is enabled for the administrator mailbox, you can complete voice mail tasks by issuing spoken commands or by pressing dialpad digits.

Accessing the Voice Mail Administrator’s Mailbox
— Call the voice mail access number.
— Say “Login” or press #.
— Say or enter the administrator mailbox number. Then enter the password, and press #.
— Say “Nine” or press 9 to access the voice mail administrator menu.

To Record a Broadcast Message:
— Access the voice mail administrator’s mailbox and menu (see above).
— Say “Broadcast Message” or press 1 to send a broadcast message.
— Record the message.
— When the message is recorded, hang up or press # for more options.

To Perform Mailbox Maintenance:
— Access the voice mail administrator’s mailbox and menu (see above).
— Say “Mailbox Maintenance” or press 2 to select the Mailbox/Group List Maintenance option.
— Say or enter the mailbox, extension ID, or group list number to be accessed and follow the prompts.

To Import a Fax Document:
— Using a fax machine, obtain dial tone and access the voice mail administrator’s mailbox and menu (see above).
— Say “Import Fax” or press 4 to select the Import Fax option.
— Say or enter the fax document number.

If the document number does not already exist, say “Yes” or press # if the number is correct or say “Re-enter” or press # and re-renter the number.

If the document number already exists, say “Yes” or press # to replace the document or say “Re-enter” or press 3 to enter another number.

If you did not enter a valid number, enter a new number.
— When prompted, press START on the fax machine.
— When the fax transmission is complete, say “Continue” or press # to continue importing documents, or press # to exit.

Quick Reference Guide To System, Voice Mail, and Diagnostics Administrator Features
These are the basic instructions for the most frequently used system administrator and voice mail administrator features. For detailed information on these features, refer to the Administrator Guide.

System Administrator Features
To Place the System in Night or Day Mode:
— While on-hook, enter 9 8 6 0.

To Place a Single Node in Day or Night Mode:
— While on-hook, enter 9 8 6 1 (Enable Network Night Mode) OR enter 9 8 6 2 (Enable Network Day Mode). Then enter the desired node number.

To Set System or Network Date and Time:
— While on-hook, enter 9 8 1 0 (System Date/Time), or enter 9 8 1 1 (Network Date/Time).
— Use the dialpad buttons to enter the month, day, and year. For example, press 010305 for January 3, 2005. Or, press # to skip ahead without changing the date.
— Use the dialpad buttons to enter the time in hours and minutes. For example, enter 0900 for 9:00. Or press # twice to exit without changing the time.
— If the system is set for 12-hour display format, press # for AM or press # for PM.

To Synchronize Network Time:
While on-hook, enter 9 8 1 1

To Respond to an Alarm Message:
— When a minor alarm indication appears, write down the alarm information.
— While on-hook, clear the alarm by entering 9 8 5 0 (Clear System Alarm) OR entering 9 8 5 1 (Clear Network Alarm).
— Look up the alarm in the Administrator Guide and take the appropriate action.
To Program System Speed-Dial Numbers:

— While on-hook, enter 9 8 0 1.
— Enter the speed dial location code (000-999) or (0000-4999).

To change or program the name: Enter the desired name for the speed-dial number using one of the following methods:

In numeric mode, the dialpad buttons are used to enter numbers 0-9, the button is used for entering a hyphen, and the button is used for entering a colon.

In alphanumeric mode, dialpad buttons are used to enter the desired letters, numbers, and punctuation. The number of times a button is pressed determines which character is entered. When adjoining characters are located under the same button, press FWD to advance to the next character. Refer to the following chart to program information in alphanumeric mode. (Note that letters correspond to the letters printed on the dialpad buttons.)

To erase the current name and leave it blank: Press MUTE repeatedly until the name is erased. Then press #.

To leave the name the same: Press #.

— Enter the number (up to 48 digits) to be stored.
— Press #.

*The Japanese characters are only available if the multilingual feature is enabled and Japanese is installed as the secondary language.

**The character available depends on the software version.

Default Administrator Feature Codes

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<th>FEATURE NAME</th>
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System Diagnostics Features

System diagnostics mode allows a system administrator to perform diagnostics functions such as freezing/unfreezing the database, printing message logs, and seizing a device for troubleshooting purposes. Use diagnostics mode when the system is idle to avoid affecting system performance.

To Enable/Disable System Diagnostics Mode:

While on-hook, enter the Diagnostics Mode feature code (9900 or 9100) to enable/disable diagnostics mode. The display shows DIAGNOSTICS ON/OFF. Then enter one of the following diagnostics feature codes (9xxx).
NOTICE

This Inter-Tel® 5000 Administrator Guide is released by Inter-Tel, Inc. as a guide for system and voice mail administrators. It provides information necessary to properly administer the system.

The contents of this guide, which reflect current Inter-Tel standards, are subject to revision or change without notice. Some features or applications mentioned may require a future release and are not available in the initial release. Future product features and applications are subject to availability and cost. Some features or applications may require additional hardware and/or specific software. Software packages released after the publication of this guide will be documented in addenda to the guide or succeeding issues of the guide.

For sales, service, or technical support in the United States, contact your local authorized Inter-Tel provider.

For additional information and/or technical assistance in Europe, system administrators may contact:

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If you have any questions or comments regarding this guide or other technical documentation, contact the Inter-Tel Technical Publications Department at Tech_Pubs@inter-tel.com.

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Product Disposal Instructions

This symbol indicates that the product is classified as electrical or electronic equipment and should not be disposed of with other commercial or household waste at the end of its working life. For appropriate disposal and recycling instructions, contact your local recycling authority or Inter-Tel provider.

The Waste of Electrical and Electronic Equipment (WEEE) Directive (2002/96/EC) was established by the European Union to minimize negative impact on the environment, control hazardous substances, and curtail landfill expansion by using the best available recovery and recycling techniques.
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**FCC Regulations**

**IMPORTANT**

1. This equipment complies with Part 68 of Federal Communications Commission (FCC) rules. On the side of the equipment Inter-Tel 5000 Base Server is a label that contains, among other information, the FCC registration number and ringer equivalence number (REN) for this equipment. Customers connecting this equipment to the telephone network shall, before such connection is made, give notice to the telephone company of the particular line(s) to which such connection is to be made, and shall provide the telephone company with the following information:

   - Complies with Part 68 of the FCC Rules
   - FCC registration number: BE2MF00BCS5000
   - USOC numbers of required interface jacks (see chart on next page)
   - Service order code (SOC), as applicable (see chart on next page)
   - Facility interface code (FIC) (see chart on next page)
   - REN, as applicable (see chart on next page)

   **NOTE**
   
   The REN is used to determine the number of devices that can be connected to a single telephone line so that all the devices will ring or otherwise respond to incoming calls. Each device that can be connected to a telephone line has a REN value. When the sum of device RENs on a line exceeds the maximum REN value the line can support, usually 5.0, some or all of the devices may not ring or respond appropriately. To identify the REN for lines in a telephone service area, contact the telephone company.

   The telephone company should also be given notice upon final disconnection of this equipment from the particular line(s).

   It is also the responsibility of the customer to provide the telephone company with registration numbers of any other devices which are configured for connection to the telephone network.

2. This equipment cannot be used on public coin service provided by the telephone company. Connection to party line service is subject to state tariffs. (Contact the state public utility commission, public service commission, or corporation commission for information.)

3. If this equipment causes harm to the telephone network, the telephone company will notify the customer in advance that service may be temporarily discontinued. But if advance notice is not practical, the telephone company will notify the customer as soon as possible. Also, the customer will be advised of the right to file a complaint with the FCC, if necessary.
4. The telephone company may make changes in its facilities, equipment, operations, or procedures which may affect the operation of this equipment. If so, the customer shall be given advance notice so that any necessary modifications can be made in order to maintain uninterrupted service.

5. If trouble is experienced with this equipment, contact a local authorized factory service representative for repairs and/or warranty information. The customer, users, and unauthorized technicians should not repair, make adjustments to, or attempt to service this equipment in any way.

6. In the event of trouble with the telephone line(s), this equipment must be disconnected from the telephone line(s). If trouble ceases, the equipment must be repaired by an authorized factory service representative. If the trouble continues to occur with the equipment disconnected, the telephone company should be notified that they have a problem. If this is the case, repairs or adjustments made by the telephone company will be made at their expense.

7. Allowing this equipment to be operated in such a manner as to not provide proper answer supervision signaling is in violation of Part 68 of FCC rules. This equipment returns answer supervision signals to the public telephone network when: answered by the called station, answered by the attendant, routed to a recorded announcement that can be administered by the equipment user, and/or routed to a dial prompt. This equipment also returns answer supervision on all DID calls forwarded back to the public telephone network. Permissible exceptions are: a call is unanswered, busy tone is received, or reorder tone is received.

8. This equipment is capable of providing users access to interstate providers of operator services through the use of equal access codes. Failure to provide equal access capabilities is a violation of the Telephone Operator Consumer Services Improvement Act of 1990 and Part 68 of the FCC Rules.

<table>
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<tr>
<th>Type of Port Interface</th>
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</table>
This equipment generates and uses radio frequency (RF) energy and if not installed and used properly, that is, in strict accordance with the manufacturer’s instructions, may cause interference to radio and television reception. This equipment has been tested and found to comply with the limits for a Class A digital device, pursuant to Part 15 of the FCC Rules. These limits are designed to provide reasonable protection against harmful interference when the equipment is operated in a commercial environment. This equipment generates, uses, and can radiate radio frequency energy and, if not installed and used in accordance with the instruction manual, may cause harmful interference to radio communications. Operation of this equipment in a residential area is likely to cause harmful interference in which case the user will be required to correct the interference at his own expense. The user is encouraged to try to correct the interference by one or more of the following measures:

- Reorient the receiving antenna.
- Relocate the equipment Inter-Tel 5000 Base Server with respect to the receiver.
- Check that the equipment Inter-Tel 5000 Base Server and receiver are not on the same circuit; the equipment must be powered from an isolated, dedicated AC outlet.
- If necessary, the user should consult the dealer or an experienced radio/television technician for additional suggestions. The user may find the following booklet prepared by the FCC helpful: “How to Identify and Resolve Radio-TV Interference Problems.”

This booklet is available from the U.S. Government Printing Office, Washington, D.C. 20402, Stock No. 004-000-00398-5.

If radio frequency interference (RFI) problems persist, contact Inter-Tel Customer Support.

The Inter-Tel® 5000 Network Communications Solutions is now product safety certified by Canadian Standards Association (CSA) for use in both the United States and Canada.
IMPORTANT SAFETY INSTRUCTIONS

NOTICE

The following safety information is reprinted from UL 1459. When using your telephone equipment, basic safety precautions should always be followed to reduce the risk of fire, electrical shock, and injury to persons.

1. Read and understand all instructions.
2. Follow all warnings and instructions marked on the product.
3. Unplug this product from the wall outlet before cleaning. Do not use liquid cleaners or aerosol cleaners. Use a damp cloth for cleaning.
4. Do not use this product near water (for example, in a wet basement).
5. Do not place this product on an unstable cart, stand, or table. The product may fall, causing serious damage to the product.
6. Slots and openings in the Inter-Tel 5000 Base Server and the back or bottom are provided for ventilation, to protect it from overheating; these openings must not be blocked or covered. This product should never be placed near or over a radiator or heat register. This product should not be placed in a built-in installation unless proper ventilation is provided.
7. This product should be operated only from the type of power source indicated in the manual. If you are not sure of the type of power source to your building, consult your dealer or local power company.
8. This product is equipped with a three-wire grounding type plug, a plug having a third (grounding) pin. This plug will only fit into a grounding type power outlet. This is a safety feature. If you are unable to insert the plug into the outlet, contact your electrician to replace your obsolete outlet. Do not defeat the safety purpose of the grounding type plug.
9. Do not allow anything to rest on the power cord. Do not locate this product where the cord will be abused by persons walking on it.
10. Do not use an extension cord with this product’s AC power cord. The AC outlet for this product should not be used for any other electrical equipment.
11. Never push objects of any kind into this product through Inter-Tel 5000 Base Server slots as they may touch dangerous voltage points or short out parts that could result in a risk of fire or electric shock. Never spill liquid of any kind on the product.

12. To reduce the risk of electric shock, do not disassemble this product, but take it to a qualified serviceman when some service or repair work is required. Opening or removing covers may expose you to dangerous voltages or other risks. Incorrect reassembly can cause electric shock when the product is subsequently used.

13. Unplug this product from the wall outlet and refer servicing to qualified service personnel under the following conditions:
   - When the power supply cord or plug is damaged or frayed.
   - If liquid has been spilled into the product.
   - If the product has been exposed to rain or water.
   - If the product does not operate normally by following the operating instructions. Adjust only those controls that are covered by the operating instructions because improper adjustment of other controls may result in damage and will often require extensive work by a qualified technician to restore the product to normal operation.
   - If the product has been dropped or the Inter-Tel 5000 Base Server has been damaged.
   - If the product exhibits a distinct change in performance.

14. Avoid using a telephone (other than a cordless type) during an electrical storm. There may be a remote risk of electric shock from lightning.

15. Do not use the telephone to report a gas leak in the vicinity of the leak.

SAVE THESE INSTRUCTIONS

NOTICE

An exclamation point within a triangle, such as the one silk-screened on the front of the Inter-Tel 5000 equipment cabinet, or Inter-Tel 5000 Base Server, is intended to alert the user to a hazard that may result in bodily harm. To promote safe operation, maintenance, and servicing of equipment, pay attention to the hazards identified in the literature accompanying the product.

Be sure to read and follow all of the instructions included in this manual and other applicable literature.

NOTICE

The equipment contains no operator-accessible areas and is certified only for installation in a restricted access location, by trained personnel only. (UL 60950-1)
Network Security Statement

Although no telecommunications system or data network is entirely secure, as long as appropriate security measures are put in place and properly maintained by both the customer and the installing company, this Inter-Tel® Advanced Communications Platform architecture and its associated server-based applications are substantially secure against unauthorized access to the customer’s data network via the telecommunications system. Appropriate security measures include, but are not limited to, the proper implementation of user/administrative accounts, passwords, firewalls, Network Address Translation (NAT), access control lists, virus protection, security updates, etc., and the proper maintenance of access points/programs and their respective accounts/passwords.
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Introduction

The Inter-Tel® 5000 Administrator Guide contains information that system administrators and voice mail administrators need to know about the Inter-Tel 5000 Network Communications Solutions family of products. The guide focuses on system hardware and software features and on the use of system administrator and voice mail administrator endpoint features needed to perform typical administrator tasks and activities. Instructions for using non-administrator endpoints and the features of other endpoints appear in the user guides that come with those devices.

This chapter describes the Administrator Guide content and organization, the intended audiences, typographical conventions used to identify safety hazards and important information, related Inter-Tel, Inc. technical documents, and a list of abbreviations and acronyms.

Endpoints and Phones

This administrator guide uses the term “endpoint” to describe the devices that the Inter-Tel 5000 family of products supports over Internet Protocol. In this context, digital and IP phones are endpoints, as are data modules, fax machines, computer telephony (CT) gateways, Single-Line Adapters (SLA), IP Single-Line Adapters (IP SLA), and answering machines.

To promote clear communication and understanding, commonly accepted names of endpoints are used as appropriate. In particular, desktop IP endpoints are also referred to as IP phones. Digital endpoints are also referred to as digital phones. Wireless endpoints are also referred to as wireless phones. And, analog single line endpoints are also referred to as single line phones or single line sets.

NOTE

In this Administrator Guide and in the Database (DB) Programming application, an endpoint or phone may also be called a “station” if reference to the endpoint or phone and its environment are intended. Or, it may be called “device” in the most general sense.
Changes to Issue 2.3

Table 1 summarizes the features and Inter-Tel 5000 capabilities introduced in v2.3.

**Table 1. Inter-Tel 5000 v2.3 Feature and Capability Descriptions**

<table>
<thead>
<tr>
<th>Feature or Capability</th>
<th>Description</th>
</tr>
</thead>
<tbody>
<tr>
<td>Dual-port digital trunk module</td>
<td>The Dual T1/E1/PRI Module (T1M-2) is the central hardware feature of v2.3. The T1M-2 module with applicable licensing provides connectivity for any combination of two T1, T1/PRI or E1/PRI trunks. Up to six T1M-2 modules can be installed in the Base Server. Increased trunk capacity may impact the system administrator’s programming planning and record-keeping activities.</td>
</tr>
<tr>
<td>Expanded echo cancellation options</td>
<td>The Echo Profiles folder in DB Programming provides options for eliminating or reducing echo that may occur on a system’s physical devices, such as trunks, endpoints, modems, etc. The Echo Profiles folder contains all the default echo profiles in the system and, as appropriate, devices can be programmed with more acceptable echo profiles. These expanded options provide more options to the system or voice mail administrator trying to resolve unacceptable echo issues at endpoints and other physical devices.</td>
</tr>
</tbody>
</table>

**Intended Audiences**

This guide provides information and procedures for carrying out typical system administrator and voice mail administrator tasks, which are summarized in the following sections.

**System Administrator**

Table 2 summarizes tasks typically performed by system administrators and provides references to applicable procedures.

**Table 2. System Administrator Tasks and Procedure References**

<table>
<thead>
<tr>
<th>Task</th>
<th>Procedure Reference</th>
</tr>
</thead>
<tbody>
<tr>
<td>Place the local system in night or day mode.</td>
<td>“Placing the Local System in Day or Night Mode” on page 27</td>
</tr>
<tr>
<td>Place the other systems in the network in night or day mode.</td>
<td>“Placing Network Nodes in Day or Night Mode” on page 27</td>
</tr>
<tr>
<td>Set the date and time of the local system.</td>
<td>“Setting System Date and Time” on page 28</td>
</tr>
<tr>
<td>Set the network date and time.</td>
<td>“Setting Network Date and Time” on page 29</td>
</tr>
<tr>
<td>Synchronize clocks in the network.</td>
<td>“Synchronizing Network Time” on page 30</td>
</tr>
<tr>
<td>Make database changes.</td>
<td>“Programming the Database Using a System Administrator Endpoint” on page 30</td>
</tr>
<tr>
<td>Program System Speed Dial numbers on the local system.</td>
<td>“Programming System Speed Dial” on page 50</td>
</tr>
<tr>
<td>Receive and clear displayed system and network alarms.</td>
<td>“Managing System and Network Alarms” on page 54</td>
</tr>
</tbody>
</table>
Any Inter-Tel endpoint can be designated as a system administrator endpoint through DB Programming. However, to facilitate programming and to manage alarm messages, Inter-Tel recommends using a Model 8660 Six-Line Display IP endpoint or a Model 8560, Executive, or Professional Six-Line Display digital endpoint for this purpose. System administrator features are described in detail beginning on page 25.

### Voice Mail Administrator

Voice mail administrators can use special features that are not provided to other voice mail users. The mailbox designated as the Administrator mailbox has all of the standard subscriber features, which can be found in the appropriate user guide, plus the ability to perform the tasks summarized in Table 3.

**Table 3. Voice Mail Administrator Tasks and Procedure References**

<table>
<thead>
<tr>
<th>Task</th>
<th>Procedure Reference</th>
</tr>
</thead>
<tbody>
<tr>
<td>Record a broadcast message.</td>
<td>“Recording a Broadcast Message” on page 63</td>
</tr>
<tr>
<td>Perform mailbox and group list maintenance.</td>
<td>“Maintaining Mailboxes and Group Lists” on page 63</td>
</tr>
<tr>
<td>Create and select customized voice mail company greetings, Automated Attendant recordings, call routing announcements, and hunt group overflow and announcement station recordings.</td>
<td>“Creating Audiotex Recordings” on page 71</td>
</tr>
<tr>
<td>Import fax documents.</td>
<td>“Importing Fax Documents” on page 70</td>
</tr>
<tr>
<td>Customize voice mail prompts.</td>
<td>“Recording Custom Voice Mail Prompts” on page 73</td>
</tr>
</tbody>
</table>

Voice mail administrator procedures are described on page 63.
Chapters of the Administrator Guide

This Administrator Guide includes the following chapters:

- **Document Overview**: This chapter provides information about the document’s structure and content, associated documents, typographical methods used to call the reader’s attention to important information, and a listing of selected abbreviations and acronyms used throughout the document.

- **Product Description**: The Product Description chapter describes system processes, capacity and feature summaries, and licensing requirements.

- **Administrator Procedures**: This chapter provides procedures for performing typical system administrator and voice mail administrator tasks and activities.

- **Hardware and Endpoints**: This chapter describes the physical equipment that makes up the Inter-Tel 5000 Network Communications Solutions. Included are the Inter-Tel 5000 Base Server; the Inter-Tel 5000 Processing Server (PS-1); the Digital Expansion Interface (DEI); Processor Module (PM-1); trunk and endpoint modules; IP and digital endpoints; and, other endpoints.

- **System Features**: The System Features chapter describes the business communication features available to Inter-Tel 5000 users. The chapter identifies trunk and feature codes, endpoint and endpoint features, system management and control features, and Administrator and Attendant features.

- **Voice Processing Features**: The Voice Processing Features chapter describes Basic Voice Mail features and optional external Voice Mail options.

- **Reference**: The Reference chapter contains information that applies to any of the Inter-Tel 5000 systems.
Multicultural English References

This administrator guide supports the requirements of technicians in the United States of America (USA) and in the Inter-Tel Europe market, which includes the United Kingdom (UK). Because of this dual support, dual references are made to industry features, standards, and jargon, as appropriate throughout the document. Terms used in the Inter-Tel Europe market appear in brackets [<European equivalent>]. For the purposes of this administrator guide, British English terms are assumed to apply to other English-speaking European cultures, as well.

For example, the U.S. telecom industry refers to an audio communication line between a public switching system and a private switching system as a “central office (CO) trunk.” In the UK, this same type of line is called a “local exchange trunk.” Applying the dual-reference guideline would result in the following statements:

- During system installation, the technician connects CO trunks [local exchange trunks] to the LSM-2 or LSM-4 modules.
- To turn the Diagnostics Mode on or off at an Administrator endpoint, the system administrator enters 9900 [9100 in Europe].

However, this administrator guide does not make a similar distinction between American English and British English spellings of common words. Only American English spellings appear in this manual. For example, the word “analog” is not also spelled “analogue” inasmuch as the meaning is clear.
Hazard Identification and Supplemental Information

This guide uses standardized graphic conventions to promote safe practices, to prevent or minimize damage to equipment, to avoid the corruption of data, and to clarify the Inter-Tel, Inc. position on certain technical or business issues.

Potentially harmful situations are identified in boxes labeled with internationally recognized safety graphics according to the degree of hazard exposure. Typically, a technician is exposed to all the identified hazards, while an administrator is exposed only to data corruption hazards.

In addition the identification of safety hazards, relevant Inter-Tel, Inc. policies, important information, and notes are set apart from the flow of the text in boxes with attention-getting labels. This section explains the conventions used and their meanings.

Hazard Identification Methods

Three hazard identification methods appear in this manual, as described in the following paragraphs.

**WARNING**

A WARNING identifies a hazard that, if not heeded, could cause definite personal injury.

![WARNING]

Possible electrical shock. Use the LCD panel to shut down the system, then turn OFF the AC POWER before working on the external voice processing PC.

**CAUTION**

A CAUTION identifies a hazard that, if not heeded, could cause minor personal injury.

![CAUTION]

Possible pinch hazard. Keep fingers clear of the edges of the faceplate when inserting a trunk module into the Inter-Tel 5000 Base Server.

Supplemental Information

Notices, Notes, and Important messages provide supplemental information about the topic being discussed.

A NOTICE stipulates conditions, requirements, and/or policies relevant to the topic under discussion.

![NOTICE]

For correct operation of the system, use ONLY the compact flash memory cards available through Inter-Tel, Inc.

A NOTE conveys supplemental information related to the topic.

![NOTE]

Commercially available premoistened towels provide adequate cleaning.

An IMPORTANT message contains information that requires special attention related to the topic.

![IMPORTANT]

Make sure the Inter-Tel 5000 Base Server connects firmly to earth ground at the Ground Lug.
Product Description

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Introduction

The Inter-Tel® 5000 platform provides an enhanced IP solution for small- and medium-sized business environments. Marketed as Inter-Tel CS-5200, CS-5400, and CS-5600 Communication Servers, each system supports IP endpoints, Inter-Tel digital endpoints, Inter-Tel IP softphones, standard single line and other endpoints. Depending on optional equipment and programming options, trunking, and licensing, the Inter-Tel 5000 platform supports 25–250 IP endpoints, up to 192 digital endpoints, and up to 54 single line analog endpoints.

Technology Highlights

This guide provides detailed descriptions of system capabilities, capacities, and applications based on the following technology features:

- **Software and Hardware Foundation**: Inter-Tel 5000 systems provide call processing (CP) features, Wide Area Network (WAN) Fail Survivability, and a Power Fail Bypass relay.

- **Scalable**: In addition to three bays that can be equipped with trunk and endpoint modules, the Inter-Tel 5000 Base Server provides multiple access ports to the platform. Two loop start trunks, two single line endpoints, and external voice mail PCs can be connected, as well as Digital Expansion Interface (DEI), Music on Hold (MOH), and paging equipment.

- **Messaging**: Basic Voice Mail (BVM) is provided with each system, and additional voice mail capacity is available by installing either the optional Enterprise® Messaging (EM) system or a Voice Processing Unit (VPU). For details, see “External Voice Processing Systems” on page 22. Fax over IP (FoIP) between Inter-Tel 5000 nodes and Unified Messaging are also available.

- **Digital signal processing**: Digital signal processor (DSP) technology enables Conferencing, Dual Tone Multi-Frequency (DTMF) detection, tone generation, modem operation, and Automatic Number Identification (ANI), or Caller ID.

- **Private networking**: Any of the Inter-Tel 5000 systems can operate as a stand-alone system or function as a node in a network of up to 99 Inter-Tel Axxess or Inter-Tel 5000 nodes. For details about 99 Nodes requirements, see “Network Capability” on page 302.

- **Oversubscription**: For systems that utilize numerous IP endpoints, IP trunks, and IP networking, v2.x allows oversubscription of IP resources. With oversubscription, dynamic allocation of IP resources, and Camp On capability, a system can be programmed with more endpoints than it has IP resources available to support simultaneously. For further explanation, see “IP Resources” on page 301. The Resource Reservation Tool allows reservation of IP resources for essential functions such as Emergency/911, as well as other customer-specified devices and endpoints.

Stable and reliable performance of the Inter-Tel 5000 platform relies, in part, on maintaining a clean and controlled environment for the Inter-Tel 5000 equipment. For environmental requirements, see Table 42 on page 298.
System Configurations

The Inter-Tel 5000 platform can be configured as any of three IP-centric systems capable of private networking and IP, digital, and analog endpoint expansion. The Inter-Tel CS-5200, CS-5400, and CS-5600 systems are sold separately as self-contained, licensed kits. Each system kit includes five IP endpoint licenses, and additional licenses are required before more than five IP endpoints will operate on a system.

On any of the Inter-Tel 5000 systems, 192 digital and 54 single line analog endpoints can be installed. For details, see “Digital Capability” on page 15.

Built into the Base Server of the three systems are two single line endpoint ports and two loop start trunk ports that provide power fail bypass capability. Depending on system configuration and allowable tradeoffs, each system can support up to 2–182 trunks. A trunk is an audio path that connects the Inter-Tel 5000 platform with the public telephone network. To see a breakout of system trunk capacities by trunk type, see Table 43 on page 299.

Inter-Tel CS-5200

The Inter-Tel CS-5200 is the basic system. In the default configuration, it comes with five IP endpoint licenses and three IP networking resources.

Running v2.x and utilizing the system’s oversubscription capability, the Inter-Tel CS-5200 supports a mix of 25–75 IP endpoints, IP private networking trunks, and/or IP gateway trunks. Required interface hardware and licensing apply.

Running v2.x software with applicable licensing, the CS-5400 supports 0, 4, or 8 BVM ports. With BVM disabled, the CS-5200 provides 32 voice channels for EM or a Windows 2000-based VPU.

Inter-Tel CS-5400

The Inter-Tel CS-5400 is a CS-5200 with a Processor Expansion Card (PEC-1) attached to the Processor Module (PM-1) and additional licensing. In the default configuration, the CS-5400 comes with five IP endpoint licenses and six IP networking resources.

Running v2.x and utilizing the system’s oversubscription capability, the Inter-Tel CS-5400 supports a mix of 110–175 IP endpoints, IP private networking trunks, and/or IP gateway trunks. Required interface hardware and licensing apply.

Running v2.x software with applicable licensing, the CS-5400 supports 0, 4, or 8 BVM ports. With BVM disabled, the CS-5400 provides 32 voice channels for EM or a Windows 2000-based VPU.

Inter-Tel CS-5600

Version 2.0 introduced the Inter-Tel CS-5600 system, which consists of an Inter-Tel CS-5400-configured Base Server connected to the same LAN as a Processing Server (PS-1). Additional network environment requirements apply. For details, see your Inter-Tel equipment service provider.

Running v2.x and utilizing the platform’s oversubscription capability, the Inter-Tel CS-5600 supports a mix of 110–250 IP endpoints, IP private networking trunks, and/or IP gateway trunks. Required interface hardware and licensing apply. The CS-5600 provides 16 voice channels for Basic Voice Mail and 32 voice channels for EM or a Windows 2000-based VPU.

Version 2.1 and later supports the eight-port Single Line Module (SLM-8) in addition to the four-port SLM-4 that can be installed in the Inter-Tel 5000 Base Server. Having the same form-factor as the Digital Endpoint Module (DEM-16), the SLM-8 fits in the Digital Expansion Interface (DEI) unit.
With v2.1 and later, the CS-5600 Processing Server (PS-1) also supports 16-port BVM and handles call processing chores for the entire system. It controls and communicates with the Base Server across the customer’s LAN. For more details about BVM, see “Basic Voice Mail” on page 287.

With the Inter-Tel CS-5600 platform, a separate Processing Server (PS-1) supports 0, 4, 8, or 16 ports of Basic Voice Mail and handles call processing chores for the entire system. It controls and communicates with the Base Server across the customer’s local area network (LAN).

**Inter-Tel 5000 Base Server**

The Inter-Tel 5000 Base Server is the foundation of the Inter-Tel 5000 Network Communications Solutions family of products. This section summarizes the technology of the Base Server and identifies system interfaces to network and customer environments.

Built on 32-bit architecture, the Inter-Tel 5000 Base Server provides a robust IP communication solution. With appropriate licensing, the Inter-Tel 5000 delivers valuable communication assets, grouped below as Hardware and Software Foundations, Networking and System Features, and Endpoints and Applications.

**Hardware and Software Foundations**

The following elements form the essential foundation of the Inter-Tel 5000 platform:

- Linux® operating system.
- Compact flash-type Inter-Tel memory card for storing system applications, customer database, BVM prompts, and voice mail messages.
- Backing up of voice mail messages to USB flash drive or Windows-based PC.
- 1U cases for housing the Inter-Tel 5000 Base Server, the Digital Expansion Interface (DEI) unit, and the CS-5600 Processing Server (PS-1).
- USB security key to ensure valid system licensing.
- Open Architecture Interface (OAI).
- Connectivity to optional external voice processing systems.
- Support for Loop Start, T1, T1/Primary Rate Interface (PRI), E1/PRI, Basic Rate Interface (BRM-S) trunks in Europe, and IP gateway (SIP) trunks.
- Messaging, including Basic Voice Mail and Unified Messaging.
- Power Fail Bypass circuitry.

**NOTE**

**Power Fail Bypass:** If power to the system is interrupted, an internal electrical relay automatically connects the two single line circuits with the two loop start trunks to provide direct connection to CO [local exchange] trunk dial tone. Under normal operating conditions, the circuits can be programmed and used the same as any loop start [local exchange] trunks or single line circuits. However, when system power fails, the circuits automatically connect, regardless of any programming that may have been done.
Networking and System Features

The Inter-Tel 5000 platform supports the following networking and system-wide features:

- Networking over IP.
- Networking over T1 and E1 spans.
- Can synchronize network time with an atomic clock.
- Wide Area Network (WAN) Fail Survivability.
- Propagation of Caller ID [CLID] information to the Public Switched Telephone Network (PSTN) across Integrated Services Digital Network (ISDN) circuits.
- Can perform digital signal processor (DSP) firmware updates, manual or automatic echo suppression, audio trace diagnostics, a CP software watchdog, and Web page theme background colors.
- Multilingual Support.
- Music-On-Hold.
- Paging.
- Fax over IP (FoIP)—T.38 (Inter-Tel 5000-to-Inter-Tel 5000 nodes only).

Endpoints and Applications

For system and voice mail administrators and for end users, the Inter-Tel 5000 platform provides the following endpoints and applications:

- Supports IP, SIP, Multi-Protocol, Wireless IP, Multimedia Touch Screen, Inter-Tel digital, and analog single line endpoints.
- Supports between 25 and 250 IP endpoints, depending on system configuration.
- Supports up to 96 digital endpoints, depending on system configuration.
- Supports up to four DSS/BLF units on a multi-line digital endpoint and up to 12 DSS/BLF units on a system.
- Supports up to 54 single line analog endpoints.
- Programmable Call Processing (CP) features.
- Web access to basic system information and administrative oversight of Base Server and Processing Server (PS-1) configurations through the Administrative Web Session page(s).
- Caller ID [CLID in Europe] transmission for single line endpoints.
- Directory Look-Up.
- Hunt Groups.
- Automatic and Uniform Call Distribution.

Processing Server (PS-1)

The Processing Server (PS-1) is a Linux® server that hosts the call processing function of the Inter-Tel CS-5600 system. The PS-1 increases processing capability to support additional system capacity and application loading. The PS-1 does not supply additional IP resources. Rather, it connects to a single Base Server for endpoint, trunk, IP, and system-DSP resource support. Via a LAN connection, the PS-1 connects to the Inter-Tel 5000 Base Server configured as a CS-5400.
Digital Capability

The addition of one or two Digital Equipment Interface (DEI) units to the Inter-Tel 5000 Base Server adds digital capability to the Inter-Tel 5000 platform and provides the same call processing functionality available to IP endpoints. The following components can be installed to provide the digital capability for your system:

- **Digital Expansion Interface Unit**: Each DEI provides three bays that accept Digital Endpoint Modules (DEM-16) and Single Line Modules (SLM-8). Each DEM-16 supports 16 digital Inter-Tel endpoints or Single-Line Adapters (SLA), providing for connection of up to 48 Inter-Tel digital endpoints or SLAs. Each SLM-8 supports eight analog single line endpoints, and each SLA supports one or two analog single line endpoints. The status of whether an Inter-Tel 5000 platform is Equipped or Not Equipped with one or two licensed DEIs appears on the Administrative Web Session (AWS).

- **Digital Endpoint Module**: Each DEM-16 provides 16 circuits for connecting Executive, Professional, Standard, and Associate Display endpoints, Axxess and Eclipse Basic Digital endpoints, Model 8500, 8520, and 8560 endpoints, or Single-Line Adapters (SLA) to the system.

- **Single-Line Adapter**: Installed on a DEM-16 circuit, a Single-Line Adapter (SLA) is used for connecting two AC-ringing, single line DTMF or pulse-tone endpoints, playback devices, and/or off-premises extensions.

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**IMPORTANT**

SLAs do not support visual Message Waiting applications. If single line sets with Message Waiting lamps are installed on SLAs, the lamps will not function. However, single line sets can be configured in DB Programming to receive audible Message Waiting tones.

---

**NOTICE**

Possible system degradation. For optimal system performance, do not use SLA circuits for Voice Mail and Automated Attendant functions. Rather, the Voice Processing PC and applications software should be used for these purposes.

---

- **Data Port Modules**: The Personal Computer Data Port Module (PCDPM) and Modem Data Port Module (MDPM) are connected to appropriate digital endpoints to provide hard-wire paths necessary to connect Digital Station Select (DSS)/Busy Lamp Field (BLF) units. Contrary to their names, however, neither device supports serial data.
  - The PCDPM can be installed on Standard, Associate, Executive, and Professional Display endpoints or on Models 8520 or 8560 endpoints to gain Off-Hook Voice Announce (OVHA) receive capability or to install an optional Modem Data Port Module (MDPM).

**NOTE**

The PCDPM supports audio only when used in conjunction with the Inter-Tel 5000 DEI product. In this version of software, the serial port is not functional on the PCDPM, and it does not transmit data, even though the name of the device suggests that it does.

  - The Modem Data Port Module (MDPM) can be installed on Standard Display, Executive Display, Model 8520, or Model 8560 endpoints for attaching a modem-equipped data device or a single line endpoint. The MDPM requires a PCDPM. The Model 8416 Mini-DSS connects to Model 8520 or Model 8560 digital endpoints without PCDPM and MDPM devices. The only programming required for the Mini-DSS is a keymap.
Trunks and Endpoints

The Inter-Tel 5000 platform supports the following trunks, spans, gateways, and endpoints:

- **Central Office (CO) [Local Exchange] Trunks**: Up to 12 loop start [local exchange] trunks can be installed in increments of four trunks per LSM-4 module. In addition, two loop start trunks can be installed in the RJ-14 port built in on the Inter-Tel 5000 Base Server.

  - **NOTE**: The two-port Loop Start Module (LSM-2) is no longer being manufactured, but the module will be supported until phased out of use at customer sites.

- **Spans**: Up to six T1, T1/PRI, or E1/PRI spans can be connected through the Inter-Tel 5000 Base Server. The T1/E1/PRI module (T1M) supports a single span, and the Dual T1/E1/PRI module (T1M-2) supports two spans of the three types.

  - **NOTE**: The single-port T1/E1/PRI Module (T1M) is in an end-of-life cycle, but the module will be supported until sold out and phased out of use at customer sites.

  - **T1**: In Database (DB) Programming each of the 24 channels of a channelized T1 span can be enabled as any of the following types of channels or circuits:
    - Loop Start
    - Ground Start
    - DID (Direct Inward Dial) ([in Europe, DDI (Direct Dial Inward)])
    - E&M (Ear and Mouth/2-way)
    - OPX (Off Premises Extension)

  - **T1/PRI**: A T1/PRI span provides 23 information-bearing (B) channels and one data (D) channel for signaling and control.

  - **E1/PRI**: An E1/PRI span provides 30 B channels and one D channel.

- **IP Devices**: The Inter-Tel 5000 platform supports Inter-Tel IP endpoints, digital endpoints, and IP Single-Line Adapter (IP-SLA) devices. The Internet Protocol Resource Application (IPRA) supplies IP functionality for the system and communicates to the endpoints and gateways across 100Base-T Ethernet. The IPRA supports the Model 8601 Pocket PC SIP softphone, which is described in the following paragraph, but does not support Inter-Tel legacy IP SoftPhones.

- **Model 8602 Softphone Application**: Model 8602 application software is distributed separately from the system software. Performance and interoperability is equivalent to hard IP endpoints.

- **Pocket PC SoftPhone**: Model 8601 is a SIP softphone that runs on selected Pocket PC 2002 (or later) Personal Digital Assistant (PDA) platforms.

- **IP Gateway Trunks**: The Inter-Tel 5000 platform supports Media Gateway Control Protocol (MGCP) and Session Initiation Protocol (SIP) gateway trunks.

- **Multi-Protocol IP Endpoints**: A multi-protocol endpoint operates either in Inter-Tel Proprietary (ITP) protocol mode or in SIP mode. A SIP server is required for SIP mode operation. Multi-protocol endpoints include Models 8600, 8620, 8622, 8662, and 8690. Multi-protocol endpoints require software licenses.

- **Multi-Protocol IP Endpoints**: Digital endpoints require DEI and DEM-16 equipment, but licenses are not required. See “Digital Endpoints” on page 97.
- **Mini-DSS**: The Inter-Tel 5000 platform supports a Mini-DSS Unit that provides one-button access to 16 extension numbers and/or feature buttons/resources.

- **Direct Station Select (DSS)/Busy Lamp Field (BLF)**: The Inter-Tel 5000 platform supports two 60-button DSS/BLF Units and one 50-button model. Installation requires DEI and DEM-16 equipment and data port modules. See “Digital Capability” on page 15.

- **Wireless Endpoints**: Models 8664, 8665, and 8668 are wireless, compact IP phones that operate in the ITP mode.

- **Single Line Endpoints**: In addition to the two circuits manufactured into the Inter-Tel 5000 Base Server, single line endpoints can be installed with IP-SLA devices, Single-Line Adapters, and four-port and eight-port Single Line Modules (SLM-4s and -8s).

### System Licensing Requirements

Software licensing controls device capacity and feature availability on the Inter-Tel 5000 platform. Licensing is associated with a system by means of a security key that inserts into a USB slot located on the front of the unit chassis.

Inter-Tel 5000 systems are delivered without software licensing. Licenses are generated and downloaded from the Inter-Tel License Management Center (LMC) Web site and then loaded onto the platform before installing the system. By default, unlicensed Inter-Tel 5000 platforms can be programmed with the capabilities of fully provisioned systems. However, if the applicable licenses are not provided and authenticated within four hours, the system invokes a reset.

If the system does not have a software license, Alarm 125 appears on the LCD panel on the Inter-Tel 5000 Base Server and on the administrator endpoint display. This alarm is generated every five minutes, and the system resets every four hours until the license issue is resolved.

### USB Security Key

The USB security key uniquely identifies a system by storing an Inter-Tel Master License. Only the specific software license that has been digitally signed with the Master License data is allowed to control the system. The key stores the essential components of the software licensing plan.

The security key also ensures licensing integrity. When a system detects removal of its key, the system disables audio for subsequent calls. When the audio is disabled, the system posts applicable alarms to the LCD panel on the front of the Base Server and to the LCD on the administrator endpoint. Alarm 122 indicates that the key is missing, while Alarm 125 identifies the number of minutes before the system initiates a reset.

When the key is replaced and the correct serial number is detected by the system, the audio is enabled and the reset countdown is aborted.

If the system powers up without the USB security key in place, the technician must reload the software license through DB Programming.

---

**NOTE**

Inter-Tel recommends provisioning each system with a display endpoint to allow an administrator to program selected system data, act as the Attendant for unsupervised CO recalls, receive system alarm messages, etc.
USB Security Key for the PS-1 Unit

The Inter-Tel CS-5600 system requires that a USB security key reside in the PS-1 unit. However, the Base Server does not need a USB security key. The administrator must upload the appropriate software license to the PS-1.

System Alarms for Insufficient Licensing

Attempts to activate an unlicensed DEI unit will trigger Alarm 130 and the unit will not come online. For details, see “Managing System and Network Alarms” on page 54. Likewise, attempting to use IP endpoints programmed in DB Programming without having appropriate IP endpoint licenses will trigger Alarm 127, and the endpoint will not come online. Inter-Tel recommends purchasing the appropriate IP endpoint license for each IP endpoint installed on the system.

System Capabilities

The Inter-Tel 5000 platform supports the telecommunication equipment required to provide seamless integration into the Public Switched Telephone Network (PSTN) and the Internet. For additional details on system capabilities, see the Reference section, beginning on page 298.

Oversubscription

With software v2.x, oversubscription allows the Inter-Tel 5000 platform to be programmed with more endpoints than there are IP resources available. This capability is useful for systems that support numerous IP endpoints and IP trunks and that utilize IP networking. If all IP resources are in use, subsequent demands for IP resources are placed in a Camp On state until IP resources become available.

Endpoint Displays

Depending on the configuration of reserved IP resources, a few scenarios may produce an unavailable IP resource condition. When system configuration results in an unavailable resource, a warning appears on display endpoints, and the user hears a beep every 5 seconds. On systems running v2.x software, the following two messages related to reserving resources may appear on display IP endpoints:

- For calls to BVM, if both reserved BVM resources and shared resources are unavailable, the calling party endpoint displays the BVM EXT IS BUSY message.
- For calls to an endpoint, if resources are not available the call is placed in a Camp On state, and the WAITING FOR RESOURCES message appears until resources become available.

Network Capability

Two or more Inter-Tel advanced communication platforms, Inter-Tel 5000 and/or Axxess®, can be connected with T1/E1/PRI network spans to form a network. To the user, the network appears as one integrated system. With few exceptions, the user can perform all of the functions across the network that can be performed within a single system or node.

For additional details on network capability, see the Reference section, beginning on page 302.
Voice Processing System Application Options

Inter-Tel voice processing systems support several voice processing system applications. Depending on your site’s voice processing system, equipment, and licensing, you may have access to one or more of the following voice processing system features.

Automated Attendant

Automated Attendant is a programmable feature that provides automated call answering services. Calls can transfer, forward, or directly ring in to Automated Attendant. When Automated Attendant answers a call, it plays a recording that gives dialing instructions. After hearing the recording, or at any time while it is playing, the caller may dial an extension or mailbox number. For information about the Automated Attendant feature, see page 265.

Automated Attendant Recall Destination

If a call that is transferred by Automated Attendant is not answered before the Transfer Voice Processor timer expires, the call recalls the Automated Attendant Recall Destination. The Recall Destination announces that the endpoint is unavailable and allows the caller to leave a message if the endpoint has an associated mailbox, or to dial another extension. For information about the Automated Attendant Recall Destination, see page 267.

Automatic Fax Detection

With Automatic Fax Detection, Call Routing Announcement (CRA) applications and mailboxes can be programmed to automatically route incoming fax calls to a specified extension or to an e-mail address. For information about the Automatic Fax Detection, see page 270.

Automatic Speech Recognition

The Enterprise Messaging (EM) system supports Automatic Speech Recognition (ASR). ASR can be enabled on many levels within the voice processing system. Using EM, subscribers can issue spoken commands to navigate the voice mail menus. Subscribers and nonsubscribers can also use spoken commands to search the voice mail directory and to navigate Automated Attendant menus and CRA paths. When ASR resources are not available, subscribers and nonsubscribers can also perform these tasks by pressing the appropriate dialpad digits when prompted by the system. For information about ASR, see page 271.

Call Logging

The Call Logging feature allows users of Inter-Tel digital and IP endpoints and Inter-Tel IP softphone applications to log the following types of calls on their respective endpoints:

- Missed Calls
- Received Calls
- Dialed Calls

The Call Logging feature helps you see who called when you were away and makes it easy to redial those people. Non-display and single line endpoints do not support this feature. Six-line display endpoints are recommended for field visibility and ease of use.
Call Routing Announcement

The Call Routing Announcement (CRA) application can be used in place of a playback device and is useful for programming Hunt Group announcement and overflow stations. When called, the CRA application plays a recording and then hangs up.

The CRA application uses digit translation to help calling parties reach their intended destinations. Digit translation allows the caller to press a single digit to access a mailbox, a Fax-on-Demand function, a station, or a Hunt Group. Digit translation can be programmed for each digit from 0 through 9, #, and *, plus a Timeout option that applies if the caller does not enter a digit when prompted. If programmed, each digit could lead the caller to a “digit translation node” that has its own set of digit translation values. This layered CRA digit translation creates a “tree” of programmable digit translation nodes. For more detailed information about CRA, see page 272.

Directory Services

Directory Services provide callers with a list of mailboxes and extension IDs that can be accessed through Voice Mail or the Automated Attendant. The Directory feature is available in each of the voice processing systems, but varies slightly in the EM system. For information about directories, see page 275. For information about using directories in EM, see page 278.

E-Mail Reader

E-mail Reader is a licensable feature available with the EM voice processing system. It provides users with advanced mailbox capabilities through Text-To-Speech (TTS) technology. When E-mail Reader is enabled for a mailbox, users can access voice mail, e-mail, and fax messages by logging in to their mailboxes. E-mail Reader announces the message envelope and can also “read” the text in the body of e-mail messages. E-mail Reader users also have access to additional message options, which vary based on the media type. For information about E-mail Reader, see page 282.

Fax-On-Demand

The Fax-On-Demand feature is a specially programmed Call Routing Announcement application that uses digit translation to allow callers to select the documents they want to have faxed to them. Callers can use a DTMF endpoint to request one or more documents from the company’s fax library, the library that is also used in the Unified Messaging faxing feature. When the request is completed, the voice processing system places a call to the caller’s fax machine to deliver the requested documents. For information about Fax-On-Demand, see page 282.

Intelligent Directory Search

The Intelligent Directory Search (IDS) feature simplifies searching for entries in a directory by significantly reducing the number of keystrokes required to find a match. IDS requires v1.2 or later. For information about IDS, see page 228.

Record-A-Call

This feature allows an endpoint user to record a call-in-progress as a voice mailbox message. When a endpoint user enters the Record-A-Call feature code, the system places a call to the endpoint’s assigned Record-A-Call application. When the application answers, the system sets up a conference call with the Record-A-Call mailbox. If programmed, the mailbox plays a greeting to indicate that the recording is in progress. For information about the Record-A-Call feature, see page 284.
Scheduled Time-based Application Routing

Scheduled Time-based Application Routing (STAR) enhances the programmability of the voice mail application greetings. Applications can be programmed to play alternative greetings for holidays and weekends. A STAR application is a table of up to 20 entries that serves as a “routing table.” STAR tells the voice processing system which application to use to respond to incoming calls. Based on day-and-time information defined in the table, the identified applications are programmed to play the greetings and STAR routes the call to the predetermined application. For information about STAR, see page 285.

Station Message Detail Recording Information Storage

Station Message Detail Recording (SMDR) information is stored only on the VPU voice processing system hard disk, then processed using call record sorting software. BVM and EM do not support SMDR information storage.

Voice Mail and Message Notification Retrieval

Other than messages that are sent through the Message Notification/Retrieval application, Voice Mail handles all voice mail calls directed to it. If your system is using EM and E-Mail Reader is enabled for the mailbox, users have access to advanced messaging features using their advanced mailbox. With Voice Mail, callers hear the main company greeting, followed by a menu of available options. Message Notification/Retrieval provides voice mail message notification and quick mailbox access. For information about Voice Mail and Message Notification Retrieval, see page 288.

Unified Messaging

The Unified Messaging (UM) feature links the voice processing system with e-mail and/or provides outgoing fax capabilities. Depending on which voice processing system you are using, you may have UM version 2.2, 2.3, or Open Standards Edition (OSE). For information about UM, see page 294.

Basic Voice Mail

Basic Voice Mail (BVM) is a built-in capability of the Inter-Tel 5000 platform that provides voice mail and messaging services. The Inter-Tel CS-5200 and CS-5400 systems support up to eight BVM ports, and the Inter-Tel CS-5600 supports up to 16 BVM ports. BVM does not support Fax-On-Demand or Unified Messaging. For details about the BVM feature, see “Basic Voice Mail” on page 287. For BVM capacities, see Table 47 on page 305.

BVM can be replaced with an external messaging system such EM. EM systems can be installed on any or all nodes in the network to allow a caller to leave a message on a local voice processing system for a mailbox located on another voice processing system in the network.
BVM Forward to E-Mail

This feature enables BVM to forward voice mail messages as a `.wav` file attachments to e-mail messages.

Features provide the following benefits:

- Allows each mailbox to be configured with a different E-Mail address.
- Operates with an industry standard SMTP mail server.
- Each mailbox can be designated to store, forward, or store-and-forward messages.

**NOTE** If both Remote Messaging and Forward to E-Mail features are enabled, the Forward and Copy option must be selected. If the Forward Only option is selected, the user will *not* receive Remote Messaging notifications.

External Voice Processing Systems

With BVM disabled, any of the Inter-Tel 5000 systems will support either the Voice Processing Unit (VPU) and the Enterprise Messaging (EM) system.

**Enterprise Messaging**

Enterprise Messaging (EM) is an optional external system that unifies communication and messaging by providing access to voice mail, fax, and e-mail messages in users mailboxes. In addition, EM provides ASR capabilities and built-in Unified Messaging features. EM runs on the Windows 2000 operating system.

**Voice Processing Unit**

The Voice Processing Unit (VPU) is an optional external system that provides integrated voice processing. Integrated features include Voice Mail, Automated Attendant, Hunt Group announcement and overflow, optional Fax-On-Demand, buffered SMDR, Record-A-Call, and others. The VPU runs on either the Windows NT® or Windows 2000 operating systems.
Voice Mail Networking

A voice processing system can be installed on any or all nodes in the user’s network. These voice processing systems can also be networked together to allow a caller to leave a message on the local voice processing system for a mailbox located on another voice processing system in the network.

Maximum capacities of networked voice processing systems are listed in Table 4.

**Table 4. Network Capacities**

<table>
<thead>
<tr>
<th>Feature/Devices</th>
<th>Capacity</th>
</tr>
</thead>
<tbody>
<tr>
<td>Voice Processing Systems per network</td>
<td>100(^1)</td>
</tr>
<tr>
<td>Local or Off-Node Mailboxes/Extension IDs per Voice Processing System node</td>
<td>10,000(^2)</td>
</tr>
<tr>
<td>System Audio Interface Ports per node</td>
<td>40</td>
</tr>
<tr>
<td>Applications per node</td>
<td>150</td>
</tr>
<tr>
<td>Group Lists per node</td>
<td>1000</td>
</tr>
<tr>
<td>Members per group list</td>
<td>1500</td>
</tr>
<tr>
<td>Remote Message Notification Numbers per mailbox</td>
<td>18</td>
</tr>
<tr>
<td>Audiotex Recordings per node</td>
<td>500</td>
</tr>
<tr>
<td>Message Notification/Retrieval Applications per node</td>
<td>1</td>
</tr>
<tr>
<td>System Passwords per node</td>
<td>4</td>
</tr>
<tr>
<td>Time Slot Groups per node</td>
<td>10</td>
</tr>
</tbody>
</table>

1. The maximum number of nodes supported by the software. System traffic may limit the actual number of nodes that can be supported without affecting system performance.

2. The maximum number of mailboxes supported by the NT-based software.
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Introduction

When programming your site’s database, the Inter-Tel-certified technician can program an endpoint’s extension as both a system administrator and a voice mail administrator. Sometimes the two jobs are handled by the same person. You can use an endpoint that has been designated as an administrator endpoint to perform certain programming tasks for the call processing system and/or the voice mail system. In addition, the system administrator endpoint shows system and network alarm messages.

This section provides you with the instructions for performing system administrator and voice mail administrator tasks. The information is presented as follows:

- System Administrator Procedures begin on page 27.
- Voice Mail Administrator Procedures begin on page 63.

A Quick Reference Guide containing instructions for frequently performed procedures is located in the front of this guide.

As a system administrator, you must know which languages are considered Primary and Secondary for the system. The platform supports American English, British English, Spanish, and Japanese prompts and displays.

System Administrator Endpoint Designation

Any Inter-Tel endpoint can be designated as a system administrator endpoint through Database (DB) Programming. However, to facilitate programming and to manage system Alarm messages, Inter-Tel recommends using a six-line display endpoint for this purpose, such as the Model 8660 IP endpoint or Model 8560 digital endpoint.

A single line endpoint cannot be used as an administrator endpoint.

If a user of a non-administrator display endpoint attempts to execute administrator features, the user hears fast-busy reorder tones, and the display shows CANNOT ACCESS RESERVED FEATURE.

Scheduled Delayed Major Resets

A delayed major reset is sometimes needed when programming the database using the administrator endpoint. If a major reset is required, the message, ENTER SYS RESET OPTION appears on the endpoint display. You can select to perform an immediate major reset when this message is shown, or you can schedule a delayed major reset. The delayed major reset option allows you to schedule a reset to occur when the system is not in use.

The following programming changes do not require a major reset:

- Changing extensions
- Swapping extensions
- Creating off-node devices
- Changing ACD Agent IDs
- Adding and removing extensions from Extension Lists
- Equipping and unequipping devices
System Administrator Procedures

With a system administrator endpoint, you can perform the following procedures:

- Place the local node in night or day mode
- Place other network nodes in night or day mode
- Set the date and time of the local node
- Set the network date and time
- Synchronize clocks in the network
- Make database changes (see page 30 for a list of programming areas)
- Enable, disable, and reset local and off-node call processing application modems
- Program System Speed Dial numbers on the local node
- Receive and clear displayed system and network alarms
- Use diagnostic mode features:
  - Freeze and unfreeze the database history for the local node or any node in the network using programmed freeze zones
  - Print error logs
  - Seize specific devices for troubleshooting purposes

Placing the Local System in Day or Night Mode

You can use a system administrator endpoint to place the local node in day or night mode. The day or night mode selects preprogrammed database lists to determine trunk access, toll restriction, emergency outgoing access, etc. for the endpoints in the system. Active calls are not affected when the Day–Night mode is changed.

To view Day–Night Mode status:

While on-hook, enter the Night Ring On/Off feature code (9860). You hear a single confirmation tone. The display shows NIGHT MODE IS ON (or OFF). If night mode was turned on, the display shows THE SYSTEM IS IN NIGHT MODE until day mode is turned on.

Placing Network Nodes in Day or Night Mode

You can use a system administrator endpoint to place one or more nodes in day or night mode. The day/night mode determines which lists the system will use for trunk access, toll restriction, etc. for the endpoints in the system. Active calls are not affected when the Day–Night mode is changed.

To set night mode on:

1. While on-hook, enter the Enable Network Night feature code (9861).
2. When prompted to enter a node number, enter the number wanted in the range 01–99. To select all nodes, enter 00.
   
   You hear a single confirmation tone. The display shows NIGHT MODE IS ON. Then the display shows NODE X IS IN NIGHT MODE until day mode is turned on.

To set day mode on:

1. While on-hook, enter the Enable Network Day feature code (9862).
2. When prompted to enter a node number, enter the node number you want.
   
   You hear a single confirmation tone. The display shows DAY MODE IS ON.
Setting System Date and Time

A system administrator can manually reset the system time or date when it is necessary, for example, when the system is defaulted or for daylight-saving time. Any administrator can change the date and time that appears on all display endpoints and in the SMDR reports. The day of the week is automatically calculated and set by the system when the date is entered. If you make a mistake while setting system date and time, press MUTE to backspace or press # or CANCEL to leave it unchanged and start over.

To set the system date and time:

1. While on-hook, enter the Set Date/Time feature code (9800). Your display shows DATE <current date>.

2. If you do not need to change the date, press 4 or ACCEPT to skip to the TIME XX:XX prompt.

3. Use the dialpad buttons to enter the month, day, and year. For example, press 020306 for February 3, 2006. When finished, the display shows TIME <current time>. If you entered the date incorrectly, the display shows INVALID DATE, and you are prompted to enter a new date.

   If using an endpoint programmed for the Japanese language, enter the date as year, month, day. For example, 060203 for February 3, 2006.

   If you do not need to change the time, press 4 or ACCEPT twice to exit. The display shows SYSTEM DATE AND TIME UPDATED.

4. Use the dialpad buttons to enter the time in hours and minutes. For example, enter 0900 for 9:00. If you entered the time incorrectly, the display shows INVALID TIME and you are prompted to enter a new time.

5. If the system is set for 12-hour display format, the display shows SELECT AM OR PM (AM=1 PM=2). Press 1 (or the AM menu button) for AM or press 2 (or the PM menu button) for PM. The display shows SYSTEM DATE AND TIME UPDATED. If you press any button other than 1 or 2, the display shows INVALID TIME and you are prompted to enter a new time.

   If using an endpoint programmed for the Japanese language, the prompts will be reversed and you will set the AM or PM before the hour and minutes.

Network Time Protocol (NTP) Support

If the Network Time Protocol (NTP) feature is enabled in DB Programming, the network date and time automatically synchronizes daily at 12:15 AM with an identified atomic clock.
Setting Network Date and Time

Occasionally, the network time or date needs to be reset, such as when the system is defaulted or when daylight-saving time/British summer time is implemented. Any system administrator can change the date and time that appears on all display endpoints and that appears in the SMDR reports in the network. The day of the week is automatically calculated and set by the system when the date is entered. If you make a mistake while setting network date and time, press MUTE to backspace or press # or CANCEL to leave it unchanged and start over.

To set the network date and time:

1. While on-hook, enter the Set Network Date and Time feature code (9810). Your display shows DATE <current date>.

2. If you do not need to change the date, press # or ACCEPT to skip to the TIME XX:XX prompt.

3. Use the dialpad buttons to enter the month, day, and year. For example, press 020306 for February 3, 2006. When finished, the display shows TIME <current time>.

   If using an endpoint programmed for Japanese, the prompts are reversed and you will set the AM or PM before the hour and minutes.

   If you entered the date incorrectly, the display shows INVALID DATE, and you are prompted to enter a new date.

   If you do not need to change the time, press # or ACCEPT twice to exit. The display shows SYSTEM DATE AND TIME UPDATED.

4. Use the dialpad buttons to enter the time in hours and minutes. For example, enter 0900 for 9:00.

   If using an endpoint programmed for Japanese, the prompts are reversed and you will set the AM or PM before the hour and minutes.

   If you entered the time incorrectly, the display shows INVALID TIME, and you are prompted to enter a new time.

5. If the node is set for 12-hour display format, the display shows SELECT AM OR PM (AM=1 PM=2). Press 1 (or the AM menu button) for AM or press 2 (or the PM menu button) for PM. The display shows SYSTEM DATE AND TIME UPDATED. If you press any button other than 1 or 2, the display shows INVALID TIME, and you are prompted to enter a new time.
Synchronizing Network Time

System administrators can synchronize the minutes past the hour across the network without changing the hour. This is useful when the nodes are in different time zones.

If the node time is off by more than 30 minutes, synchronizing the minutes may cause the hour to change. Also, network time is automatically synchronized every day at 12:30 AM (00:30), using the time setting on the node with the lowest number.

Or, if the Network Time Protocol (NTP) support feature is enabled, the system updates the network time in accordance with an atomic clock referenced to the system time zone. For more information about NTP, see page 256.

To synchronize the clocks in all nodes in the network:

While on-hook, enter the Synchronize Network Time feature code (9811). You hear a confirmation tone, and the display shows NETWORK TIME SYNCHRONIZED.

Programming the Database Using a System Administrator Endpoint

Any system administrator endpoint can program some parts of the system database. However, programming requires a display endpoint, such as an Executive Display, Professional Display or Model 8560 digital endpoint or the Model 8660 IP endpoint.

A system administrator endpoint can be used to perform the following programming:

Endpoint Options:
- Create or delete administrator stations
- Create or delete attendant stations
- Create or delete House Phones
- Assign endpoints to attendants
- Program usernames
- Program endpoint toll restriction

System Options:
- Program Do-Not-Disturb (DND) messages
- Program up to 10 passwords for the DB Programming application
- Program endpoint extensions
- Swap extensions

Trunk Options:
- Individual trunk answer supervision type, caller information, hybrid balance, signaling type (DTMF or pulse), and trunk group assignment
- Trunk group answer access, ring-in, toll restrictions, and trunk lists

A password can protect entry to the DB Programming application at the administrator endpoint. A password prevents unauthorized users from altering the system database.

IMPORTANT: Passwords are important to system security. Without sufficient password protection, the system database is vulnerable to unauthorized access.
Depending on the database changes made, the system may require a reset after programming. If so, the system will prompt you for a reset and ask if it should be done immediately or delayed. Delaying the reset may prevent interruption in service. However, if a reset is required it should be done as soon as possible to permit proper system operation.

**NOTICE**

Possible interruption of calls in progress. When the system is Reset, all calls in progress are dropped immediately.

Entering Alphanumeric Information:

When entering alphanumeric information, such as a username, reminder message, or Do-Not-Disturb message, press **USE ALPHA MODE/USE NUMERIC MODE** menu button to switch back and forth between alphanumeric and numeric mode.

- **In numeric mode**, the dialpad buttons are used to enter numbers 0–9, the pound (\#) button is used for entering a hyphen (-), and the asterisk (*) button is used for entering a colon (:). For example, \#00 would enter “1:00” in numeric mode.

- **In alphanumeric mode**, dialpad buttons are used to enter letters, numbers, and punctuation. The number of times a button is pressed determines which character is entered. For example, 533266 would enter “JEAN” in English. When adjoining characters are located under the same button, press **FWD** to advance to the next character. For example, 66 **FWD** 6667776 would enter “NORM.”

See **Table 5** to program information in alphanumeric mode. (Note that letters correspond to the letters printed on dialpad buttons.) Japanese characters are available with the standard Multilingual Support feature included in your software license. Japanese is programmed as a Primary or Secondary Language, and the administrator endpoint is set for Japanese.

**Table 5. Alphanumeric Equivalents for Dialpad Programming**

<table>
<thead>
<tr>
<th>Button</th>
<th>1</th>
<th>2</th>
<th>3</th>
<th>4</th>
<th>5</th>
<th>6</th>
<th>7</th>
<th>8</th>
<th>9</th>
<th>10</th>
<th>11</th>
</tr>
</thead>
<tbody>
<tr>
<td><strong>ENGLISH/SPANISH CHARACTERS</strong></td>
<td>-</td>
<td>&amp;</td>
<td>(</td>
<td>1</td>
<td>A</td>
<td>I</td>
<td>U</td>
<td>E</td>
<td>O</td>
<td>a</td>
<td></td>
</tr>
<tr>
<td>2</td>
<td>A</td>
<td>B</td>
<td>C</td>
<td>'</td>
<td>2</td>
<td>KA</td>
<td>KI</td>
<td>KU</td>
<td>KE</td>
<td>KO</td>
<td>i</td>
</tr>
<tr>
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<td>E</td>
<td>F</td>
<td>!</td>
<td>3</td>
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<td>SHI</td>
<td>SU</td>
<td>SE</td>
<td>SO</td>
<td>u</td>
</tr>
<tr>
<td>4</td>
<td>G</td>
<td>H</td>
<td>I</td>
<td>*</td>
<td>4</td>
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<td>CHI</td>
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<td>TE</td>
<td>TO</td>
<td>e</td>
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<tr>
<td>5</td>
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<td>5</td>
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<td>o</td>
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<td>U</td>
<td>V</td>
<td>?</td>
<td>8</td>
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<td>,</td>
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</tr>
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<td>X</td>
<td>Y</td>
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<td>RO</td>
<td>yo</td>
</tr>
<tr>
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<td>:</td>
<td>.</td>
<td>,</td>
<td>0</td>
<td>WA</td>
<td>WO</td>
<td>N</td>
<td>pa</td>
<td>ba</td>
<td>long</td>
</tr>
</tbody>
</table>

* The Japanese characters are available only if the Multilingual feature is enabled and Japanese is installed as the secondary language.
Station Programming

You can use your administrator endpoint to program the following endpoint information:

- **Create or delete administrator endpoints**: You can program any non-single line Inter-Tel endpoint as an additional administrator endpoint, or you can delete the designation of an endpoint as an administrator endpoint.

- **Create or delete attendant endpoints**: You can program any endpoint to be an attendant endpoint, or you can delete the designation of an endpoint as an attendant endpoint.

- **Create or delete House Phones**: You can program any endpoint to be a House Phone, or you can delete House Phones.

- **Assign endpoints to attendants**: You can change the assigned attendant for each endpoint.

- **Program usernames**: You can program or change the username for any endpoint.

- **Program endpoint toll restrictions**: You can program toll restriction classes of service for endpoints. Endpoint toll restrictions are described in detail on page 142.

The endpoint DB Programming process is summarized in the flowchart shown on page 35. For Program Planning Sheets, see page 309.

**NOTE** If necessary, at any time during the following procedure you can press the asterisk (*) or Speaker button to cancel programming and discard any unsaved changes.

To program endpoints:

1. While on-hook, enter the Program Database feature code (9932) [9132 in Europe].

2. If a password is required, the display shows ENTER PASSWORD. Use the dialpad to enter your 1–8 digit password and press #. (If you enter an incorrect password or do not press #, the display shows INVALID PASSWORD.)

   If a password is not required, skip this step.

3. The display shows ENTER DATABASE OPTION. (Display endpoints also show the options: ENDPOINT, SYSTEM, and TRUNK.) Press 1 or the ENDPOINT menu button.

4. The display shows ENTER ENDPOINT EXTENSION. Enter the extension number of the endpoint to be programmed using one of the following methods. (If you enter an invalid extension number, you hear reorder tones and must try again.)

   **Enter a complete number**: Enter the extension number using your dialpad. When the circuit information is displayed, press # again to continue.

   **Enter a partial number**: Enter a partial extension number, then press #, the high end of the Volume button, or the ACCEPT menu button. The display shows the extension number, username and circuit number of the endpoint that most closely matches the partial extension number. Press # or ACCEPT to program the displayed endpoint, or scroll to another endpoint, as described below.

   **Scroll through the numbers**: To scroll through the extension number list, press the high end of the Volume button or NEXT to scroll forward, or press the low end of the Volume button or PREVIOUS to scroll backward. When the correct extension number appears, press # or ACCEPT to continue.
5. The display shows ENTER ENDPOINT OPTION. Select one of the following:

- **Endpoint Flags**: This option allows you to set the administrator endpoint, attendant, and House Phone flags. To select it, press 1 or the ENDPOINT FLAGS menu button. Three flags can be programmed: Administrator, Attendant, and House Phone. However, you cannot program the Administrator flag for your own endpoint or for a single line endpoint. To program the flags, do the following:
  - To scroll to the wanted flag, press the high end of the Volume button or NEXT to scroll forward, or press the low end of the Volume button or PREVIOUS to scroll backward.
  - To enable or disable a displayed flag, press 1 or the ON menu button to enable the flag. Or, press 2 or the OFF menu button to disable the flag.
  - To save your programming when all flags are set correctly, press 3 or the ACCEPT menu button to save the new flag settings. The display shows DATABASE UPDATED and then returns to the ENTER ENDPOINT OPTION prompt.
  - To exit without saving your changes, press 3 or the CANCEL menu button to exit. The display shows NO UPDATE PERFORMED and then returns to the ENTER ENDPOINT OPTION prompt.

- **Endpoint Information**: This option allows you to select an attendant for the endpoint or program the endpoint’s username. To select it, press 2 or the ENDPOINT INFO menu button. The display shows ENTER ENDPOINT INFO OPTION. Select one of the following:
  - **Attendant**: To assign an attendant to serve this endpoint, press 1 or the ATTENDANT menu button. The display shows ENTER ATTENDANT EXTENSION. Enter the extension number you want. When the circuit information is displayed, press 4 again to return to the ENTER ENDPOINT INFO OPTION prompt. The display shows DATABASE UPDATED. (Or, to cancel your entry, press 3 or the CANCEL menu button. The display shows NO UPDATE PERFORMED.)
  - **Username**: To change the username of the endpoint, press 2 or the USERNAME menu button. The display shows ENTER USERNAME. Enter the new name as described on page 31. To cancel your entry, press 3 or the CANCEL menu button. The display shows NO UPDATE PERFORMED.

- **Toll Restriction**: This option allows you to set the endpoint class of service for day and/or night modes. To select it, press 3 or the TOLL RESTRICTION menu button. Then do the following:
  - The display shows TOLL RESTRICTION OPTION. Press 1 or the COS DAY menu button to program day mode toll restriction. Or, press 2 or the COS NIGHT menu button to program night mode toll restriction.
  - The display shows SET DAY (or NIGHT) COS XX. If this is not the COS you want to program, scroll to the correct COS by pressing the high end of the Volume button or NEXT to scroll forward, or press the low end of the Volume button or PREVIOUS to scroll backward. For more information about COS, see page 143.
The default COS numbers for the U.S. are provided in Table 6 and the default COS numbers for Europe are provided in Table 7.

**Table 6. Default Class of Service (COS) Numbers for the U.S.**

<table>
<thead>
<tr>
<th>COS</th>
<th>Description</th>
</tr>
</thead>
<tbody>
<tr>
<td>02</td>
<td>Deny Area/Office</td>
</tr>
<tr>
<td>03</td>
<td>Deny Operator</td>
</tr>
<tr>
<td>04</td>
<td>Deny Toll Access</td>
</tr>
<tr>
<td>05</td>
<td>Deny International</td>
</tr>
<tr>
<td>06</td>
<td>Deny Equal Access</td>
</tr>
<tr>
<td>07</td>
<td>Deny Local Calls</td>
</tr>
<tr>
<td>08</td>
<td>Denied Numbers</td>
</tr>
<tr>
<td>09</td>
<td>Allowed Numbers</td>
</tr>
</tbody>
</table>

**Table 7. Default Class of Service (COS) Numbers for Europe**

<table>
<thead>
<tr>
<th>COS</th>
<th>Description</th>
</tr>
</thead>
<tbody>
<tr>
<td>01</td>
<td>ARS Only</td>
</tr>
<tr>
<td>02</td>
<td>Deny Operator</td>
</tr>
<tr>
<td>03</td>
<td>Deny Toll Access</td>
</tr>
<tr>
<td>04</td>
<td>Deny International</td>
</tr>
<tr>
<td>05</td>
<td>Deny Local Calls</td>
</tr>
<tr>
<td>06</td>
<td>Denied Numbers</td>
</tr>
<tr>
<td>07</td>
<td>Allowed Numbers</td>
</tr>
</tbody>
</table>

6. When the correct COS is displayed, press 1 or the ON menu button to enable the toll restriction. Or, press 2 or the OFF menu button to disable it.

b.) If necessary, repeat the option titled “Toll Restriction” on page 33, and then to program additional COS toll restrictions, repeat step a above.

c.) Press 4 or ACCEPT to save the COS programming. The display shows DATABASE UPDATED and then returns to the TOLL RESTRICTION OPTION prompt.

d.) Press 4 again to exit to the ENTER ENDPOINT OPTION prompt.

6. When the display shows ENTER ENDPOINT OPTION, press 4 again to exit to the ENTER ENDPOINT EXTENSION prompt. You can then program another endpoint by repeating these steps or press 4 or ACCEPT once more to exit to the ENTER DATABASE OPTION prompt.

7. When finished with all programming, press 4 while the ENTER DATABASE OPTION prompt is displayed. This ends the programming session.

8. **If a system reset is required**, the display shows ENTER SYS RESET OPTION. Do one of the following:

   - **Delayed Reset**: Press 1 or the DELAYED menu button to delay the reset. The display shows DELAYED RESET SCHEDULED. The system will be reset at the pre-programmed time.
   - **Immediate Reset**: Press 2 or the IMMEDIATE menu button to reset the system now.

**NOTICE**

Possible interruption of calls in progress. When the system is Reset, all calls in progress are dropped immediately.
Figure 1. Endpoint Database Programming Flowchart

Enter Program Database feature code 9932 (9132)

Press 1 or ENDPOINT menu button

Press 1 or ENDPOINT DATABASE OPTION

Press 1 or ENDPOINT ENDPOINT EXTENSION

Type the extension number you want, and then #

Press 1 or ENDPOINT OPTION

Select flag to program

Press 1 or ON menu button to enable
Press 2 or OFF menu button to disable

Press 1 or ENDPOINT FLAGS menu button

Press 2 or ENDPOINT INFO menu button

Press 3 or TOLL_RESTRICTION menu button

Press 1 or COS DAY menu button for day mode
Press 2 or COS NIGHT menu button for night mode

IF NEEDED, SELECT COS NUMBER TO PROGRAM BY SCROLLING

Press 1 or COS ON menu button to enable COS
Press 2 or COS OFF menu button to disable COS

AT ANY LEVEL:
Press * or the Speaker button to cancel any unsaved changes and back up to exit programming.
Press # or ACCEPT to save changes and back up one display level.

INFORMATION IN THIS CHART IS SHOWN AS FOLLOWS:

YOUR INPUT

ENDPOINT DISPLAYS
System Programming

You can use your administrator endpoint to program the following system-wide information:

- **Define Do-Not-Disturb (DND) messages**: The messages for the Do-Not-Disturb feature can be reprogrammed through an administrator endpoint. Administrators can delete or change messages 01–20 to any value up to 16 characters. For feature description and DND message details, see “Do-Not-Disturb” on page 238. When the system has a programmed Primary and Secondary Language, the system has default DND messages in both languages. Available languages are American English, British English, Spanish, and Japanese. The current language of the programming endpoint determines which list is programmed. For a description of the Change Language feature, see “Multilingual Capability” on page 159. The Secondary Language translation has the same meaning as the Primary Language message. Table 8 lists the default DND messages.

<table>
<thead>
<tr>
<th></th>
<th>Default DND Messages</th>
</tr>
</thead>
<tbody>
<tr>
<td>01</td>
<td>DO-NOT-DISTURB</td>
</tr>
<tr>
<td>02</td>
<td>LEAVE A MESSAGE</td>
</tr>
<tr>
<td>03</td>
<td>IN MEETING UNTIL</td>
</tr>
<tr>
<td>04</td>
<td>IN MEETING</td>
</tr>
<tr>
<td>05</td>
<td>ON VACATION ’TIL [ON HOLIDAY ’TIL]</td>
</tr>
<tr>
<td>06</td>
<td>ON VACATION [ON HOLIDAY]</td>
</tr>
<tr>
<td>07</td>
<td>CALL ME AT</td>
</tr>
<tr>
<td>08</td>
<td>AT THE DOCTOR</td>
</tr>
<tr>
<td>09</td>
<td>ON A TRIP</td>
</tr>
<tr>
<td>10</td>
<td>ON BREAK</td>
</tr>
<tr>
<td>11</td>
<td>OUT OF TOWN ’TIL</td>
</tr>
<tr>
<td>12</td>
<td>OUT OF OFFICE</td>
</tr>
<tr>
<td>13</td>
<td>OUT UNTIL</td>
</tr>
<tr>
<td>14</td>
<td>WITH A CLIENT</td>
</tr>
<tr>
<td>15</td>
<td>WITH A GUEST</td>
</tr>
<tr>
<td>16</td>
<td>UNAVAILABLE</td>
</tr>
<tr>
<td>17</td>
<td>IN CONFERENCE</td>
</tr>
<tr>
<td>18</td>
<td>AWAY FROM DESK</td>
</tr>
<tr>
<td>19</td>
<td>GONE HOME</td>
</tr>
<tr>
<td>20</td>
<td>OUT TO LUNCH</td>
</tr>
</tbody>
</table>

When two languages are enabled and DND messages are changed, the programmer should attempt to keep the meanings for the messages in both lists the same. That is, if the Primary Language message 02 is changed to “PAGE ME,” a similar message should be programmed for the Secondary Language message 02.

- **Select an Administrator DB Programming password**: Entry to the DB Programming application at the administrator endpoints can be protected using a password. A password would prevent unauthorized users from altering the system database.

- **Define reminder messages**: System reminder messages can be changed using an administrator endpoint. For a feature description and operating instructions, see “Reminder Messages” on page 247. The messages can have up to 16 characters each. When Primary and Secondary Languages are enabled, the system has default reminder messages in both languages. The current language of the programming endpoint determines which list is programmed. For a description of the Change Language feature, see “Secondary Language Selection” on page 160. Each Secondary Language translation has the same meaning as the Primary Language message.

When two languages are enabled and reminder messages are changed, the programmer should attempt to keep the meanings for the messages in both lists the same. That is, if the Primary Language message 02 is changed to “GO TO AIRPORT,” a similar message should be programmed for the Secondary Language message 02.
• **Program new extension numbers for endpoints:** The extension number for any endpoint can be changed by an Administrator. The new extension number cannot conflict with an existing number.

• **Swap extensions:** An extension number can be relocated (swapped) to another endpoint. To swap extensions, the two affected endpoints must meet the following criteria:
  – Both devices must reside on the same node as the Administrator performing the swap.
  – Both devices must be the same type (i.e., both digital, both Inter-Tel Protocol (ITP) mode, both SIP mode, or both single line endpoints).
  – Neither device can be the administrator endpoint performing the swap.

The System-Wide DB Programming process is summarized in the flowchart shown on page 41. For Program Planning Sheets, turn to page 309.

If necessary, you can press $ or the Speaker button to cancel programming and discard any unsaved changes, at any time during the following procedure. If you want to change the Japanese DND or reminder message settings, make sure your endpoint is set in Japanese mode. For an explanation of the Change Language feature, see “Secondary Language Selection” on page 160.

**To program the system database:**

1. While on-hook, enter the Program Database feature code 9932 [9132 in Europe].

2. If a password is required, the display shows ENTER PASSWORD. Use the dialpad to enter your 1–8 digit password and press #. If you enter an incorrect password, the display shows INVALID PASSWORD.
   
   If a password is not required, skip this step.

3. The display shows ENTER DATABASE OPTION. Display endpoints show the options: ENDPOINT, SYSTEM, and TRUNK. Press 2 or the SYSTEM menu button.

4. The display shows ENTER SYSTEM OPTION. Select one of the following:
   
   • **Do-Not-Disturb Messages:** This option allows you to program the DND messages used by the endpoints. To select it, press 1 or the DND MESSAGES menu button. Then do the following:
     
     – The display shows SELECT DND MESSAGE #. Enter a message number or scroll to the message you want. To scroll to the correct message, press the Volume button or the SCROLL plus NEXT and PREVIOUS menu buttons.
     
     – When the display shows the correct DND message, enter the new message as described on page 31.
     
     – Press $ or ACCEPT to save the new message. The display shows DATABASE UPDATED and then returns to the SELECT DND MESSAGE prompt. To cancel your entry, press $ or CANCEL. The display shows NO UPDATE PERFORMED.
     
     – To program another message, scroll to the message you want and repeat these steps.
     
     – Press # again to exit to the ENTER SYSTEM OPTION prompt.
• **Password**: This option allows you to set a password that limits access to the Administrator programming feature. To select it, press \#2 or the PASSWORD menu button. Then do the following:
  
  – The display shows CHANGE PASSWORD TO. Enter a password of up to eight digits, then press \#. To erase the password and leave it blank, just press \#.
  
  – The display shows VERIFY PASSWORD. Enter the password exactly as you did in step 2 above, followed by \#. The display returns to the ENTER SYSTEM OPTION prompt. If you hear reorder/fast-busy tones and see an error message, the passwords did not match and you must start over at the CHANGE PASSWORD prompt.

• **Reminder Messages**: This option allows you to program the reminder messages used by the endpoints. To select it, press \#3 or the REMINDER MSGS menu button. Then do the following:
  
  – The display shows SELECT REMINDER MSG #. Enter a message number or scroll to the message you want. To scroll to the message, press the Volume button or the SCROLL plus NEXT and PREVIOUS menu buttons.
  
  – When the display shows the correct message, enter the new message as described on page 31.
  
  – Press \# or ACCEPT to save the new message. The display shows DATABASE UPDATED and then returns to the SELECT REMINDER MSG prompt. To cancel your entry, press \# or CANCEL. The display shows NO UPDATE PERFORMED.
  
  – To program another message, scroll to the message you want and repeat these steps.
  
  – Press \# again to exit to the ENTER SYSTEM OPTION prompt.
• **Station Extensions**: This option allows you to assign new extension numbers to stations/endpoints/phones. To select it, press 4 or the **STN EXTENSION** menu button.

**If programming an endpoint:**

- **Press 1 or CHANGE EXT.** The display shows ENTER ENDPOINT EXTENSION. Enter the extension number of the endpoint to be programmed using one of the following methods. If you enter an invalid extension number, you hear reorder tones and must try again.

  **Enter a complete number:** Enter the extension number using your dialpad. When a valid number is entered, the circuit information is displayed. Press # again to continue.

  **Enter a partial number:** Enter a partial extension number, then press #, the high end of the Volume button, or the ACCEPT menu button. The display shows the extension number, username, and circuit number of the endpoint that most closely matches the partial extension number. Press # or ACCEPT to program the displayed endpoint, or scroll to another endpoint as described below.

  **Scroll through the numbers:** To scroll through the extension number list, press the high end of the Volume button or the NEXT menu button to scroll forward, or press the low end of the Volume button or the PREVIOUS menu button to scroll backward. When the correct extension number appears, press # or ACCEPT to continue.

  - The display shows CHANGE X <number> TO EXTENSION. Enter the new extension number for the endpoint. If you enter an invalid number, you hear reorder tones and must try again. **If you entered an extension number that is already assigned**, the display shows CONFLICTING EXTENSION and you hear reorder tones. When the display returns to ENTER ENDPOINT EXTENSION, you must start over and select a new extension number.

  - Press # or ACCEPT to exit to the ENTER ENDPOINT EXTENSION prompt. The display shows UPDATING DATABASE for four seconds and then DATABASE UPDATED. (Or, to cancel your entry, press # or CANCEL. The display shows NO UPDATE PERFORMED.)

Although the system begins to update the database when you press # or ACCEPT, it may take longer than the four-second display to change the extension in the system, especially in a large or busy system.

- Press # or ACCEPT again to exit to the ENTER SYSTEM OPTION prompt. (If an extension number is displayed, press # to exit, instead of # or ACCEPT.)
If swapping endpoints:

- Press \key{2} or SWAP EXTS. The display shows ENTER ENDPOINT EXTENSION. Enter the first extension number to be swapped using one of the following methods. (If you enter an invalid extension number, you hear reorder tones and must try again.)

Enter a complete number: Enter the extension number using your dialpad. When a valid number is entered, the circuit information is displayed. Press \key{7} or ACCEPT to continue.

Enter a partial number: Enter a partial extension number, then press the \key{8} button, the high end of the Volume button, or the ACCEPT menu button. The display shows the extension number, user name and circuit number of the endpoint that most closely matches the partial extension number. Press \key{7} or ACCEPT to continue, or scroll to another endpoint as described below.

Scroll through the numbers: To scroll through the extension number list, press the high end of the Volume button or the NEXT menu button to scroll forward, or press the low end of the Volume button or the PREVIOUS menu button to scroll backward. When the correct extension number appears, press \key{7} or ACCEPT to continue.

If the feature times out while entering an extension number, the displays shows INCOMPLETE EXTENSION.

- The display shows SWAP <extension> WITH EXTENSION. Enter the second extension number to be swapped using the same methods previously described. The display shows DATABASE UPDATED.

If you cancel or terminate the feature without entering the second extension number, the system discards any changes that have not yet been saved, and the display shows NO UPDATE PERFORMED. If the entered extension is the administrator endpoint performing the swap, the same extension number as the target endpoint, or the wrong device type, the display shows CONFLICTING EXTENSION.

- When the display shows ENTER SYSTEM OPTION, press \key{7} again to exit to the ENTER DATABASE OPTION prompt.

- When finished with all programming, press \key{7} while the ENTER DATABASE OPTION prompt is displayed. This ends the programming session.

- If a system reset is required, the display shows ENTER SYS RESET OPTION. Do one of the following:
  - Delayed Reset: Press \key{1} or the DELAYED menu button to delay the reset. The display shows DELAYED RESET SCHEDULED. The system will be reset at the pre-programmed time.
  - Immediate Reset: Press \key{2} or the IMMEDIATE menu button to reset the system now.

**NOTICE**

Possible interruption of calls in progress. When the system is Reset, all calls in progress are dropped immediately.
Figure 2. System Database Programming Flowchart

Enter Program Database feature code 9932 [9132]

Press 2 or SYSTEM menu button

ENTER DATABASE OPTION

Press 1 or DND MESSAGES menu button
Press 2 or PASSWORD menu button
Press 3 or REMINDER MSGS menu button
Press 4 or STN EXTENSIONS menu button

ENTER SYSTEM OPTION

SELECT DND MESSAGE

Enter message number or select DND message to program by scrolling

(DND message) SCROLL OR EDIT

Enter the new message using alpha or numeric mode and press #

If wanted, select another message to program

CHANGE PASSWORD

Enter 0-8 digit password and press #

VERIFY PASSWORD

Enter the new password again and press #

If wanted, select another message to program

SELECT REMINDER MSG

Enter message number or select message to program by scrolling

(Reminder message) SCROLL OR EDIT

Enter the new message using alpha or numeric mode and press #

If wanted, select another message to program

ENTER ENDPOINT EXTENSION

Enter the extension number of the endpoint to be changed

CHANGE (number) SCROLL OR EDIT

Enter the new extension number for the endpoint and press #

ENTER ENDPOINT EXTENSION

Enter the extension number of the first endpoint to be swapped and press #

SWAP (extension) WITH EXTENSION

Enter the extension number of the second endpoint to be swapped and press #

AT ANY LEVEL:
Press * or the Speaker button to cancel any unsaved changes and back up to exit programming.
Press # or ACCEPT to save changes and back up one display level.

INFORMATION IN THIS CHART IS SHOWN AS FOLLOWS:

YOUR INPUT

ENDPOINT DISPLAYS
Trunk Programming

The trunk information that can be programmed by an administrator endpoint includes the following:

Individual trunks:

- **Answer Supervision type**: Answer Supervision determines whether the system should consider a call valid when it receives polarity reversal from the central office or wait for the Valid Call timer to expire. Any outside call put on hold becomes valid immediately, regardless of Answer Supervision or Valid Call timer status. There are three options available under Answer Supervision type:

  - **Polarity Reversal**: A loop reversal must be received to consider the call valid. When the first loop reversal is received, the call is made valid immediately, and the Inter-Tel endpoint display will begin call cost. When a second loop reversal is received, the system will terminate the call.

  If a second loop reversal is not received, the system will not terminate the call unless the inside party hangs up or loss-of-loop is received from the telephone company.

  - **Valid Call Timer**: After the Valid Call Timer expires, the call is validated. All polarity reversals received before and after the Valid Call Timer are ignored.

  - **Valid Call Timer with Polarity Reversal**: If a loop reversal is received before the Valid Call Timer expires, the call will be made valid immediately, and the Inter-Tel endpoint display will begin call cost. When a second loop reversal is received, the system will terminate the call. If a loop reversal is not received before the Valid Call Timer expires, the call will be made valid by the timer. If a loop reversal is received after the timer expires, the loop reversal is ignored, but the call cost is reset. If a second loop reversal is then received, the system will terminate the call.

    If a second loop reversal is not received, the system will not terminate the call unless the inside party hangs up or loss-of-loop is received from the telephone company.

- **Hybrid balance**: Except for analog trunks, hybrid balance can be set to **Normal** or **Short** to allow the installer to adjust the hybrid impedance of the trunk based on local central office conditions.

  - **Short**: The short balance is used when trunks are supplied by other equipment located within the same building, using private, copper wire termination (e.g., an OPX supplied by another PBX that does not use the public network) or when the site is located close to the central office.

  - **Normal**: The normal balance can be used by most trunks.

- **DTMF or dial pulse signaling**: Trunks can be set for DTMF or dial pulse signals.

- **Trunk group assignment**: Trunks can be assigned to trunk groups on an individual basis.

- **Busy trunk**: This feature allows you to temporarily remove a trunk from service. While the trunk is out of service, it cannot be used for placing outgoing calls. However, incoming calls on analog trunks will ring in and can be answered. Incoming calls on ISDN trunks cannot be answered. The trunk will remain out of service until the Trunk Busy Out timer expires or you place the trunk back in service.
Trunk groups:

- **Day/Night Answer Access**: You can program separate lists for endpoints with allowed-answer access in day and night modes.
- **Day/Night Ring-In**: You can program separate lists of endpoints with ring-in for day and night modes.
- **Change Toll Restrictions**: You can program toll restriction classes of service for the trunk groups. Classes of service are described in detail on page 143.
- **Trunk List**: Each trunk group has a programmed list of individual trunks.

The Trunk Database Programming process is summarized in the flowchart shown on page 48. See page 309 for Program Planning Sheets.

If necessary, at any time during the following procedure you can press the asterisk (ि) or Speaker button to cancel programming and discard any unsaved changes.

**Entering extension numbers:**

When the procedure asks you to enter an endpoint, trunk, or trunk group extension number, you can use any of the following methods. (If you enter an invalid extension number, you hear reorder tones and must try again.)

- **Enter a complete number**: Enter the extension number using your dialpad. When the circuit information is displayed, press  again to continue.
- **Enter a partial number**: Enter a partial extension number then press  the high end of the Volume button, or the  menu button. The display shows the extension number, username and circuit number of the endpoint, trunk, or trunk group that most closely matches the partial extension number. Press  or ACCEPT to accept the displayed extension, or scroll to another endpoint as described below.
- **Scroll through the numbers**: To scroll through the extension number list, press the high end of the Volume button or the NEXT menu button to scroll forward, or press the low end of the Volume button or the PREVIOUS menu button to scroll backward. When the correct extension number is displayed, press  or ACCEPT to continue.

**To program trunk information:**

1. While on-hook, enter the Program Database feature code (9932) [9132 in Europe].

2. **If a password is required**, the display shows ENTER PASSWORD. Use the dialpad to enter your 1–8 digit password and press  (If you enter an incorrect password, the display shows INVALID PASSWORD.)

   **If a password is not required**, skip this step.

3. The display shows ENTER DATABASE OPTION. Six-line display endpoints also show the following options: ENDPOINT, SYSTEM, and TRUNK. Press  or the TRUNK menu button. The display shows ENTER TRUNK OPTION.
4. **To Program an Individual Trunk:** Press 1 or the **INDIVIDUAL TRUNK** menu button. Then program trunk options as follows:

- The display shows **ENTER TRUNK EXTENSION**. Enter the extension number of the trunk to be programmed using one of the methods described on page 43.
- The display shows **INDIVIDUAL TRUNK OPTION**. Enter one of the following options. To view the last two options on a display endpoint, press the arrow (<< or >>) menu button at the bottom of the display.
  
  - **Answer Supervision:** To program the answer supervision for the trunk, press 1 or the **ANS SUPERVISION** menu button. When the display shows **ANS SUPERVISION OPTION**, press 1 or the **POLARITY REVERSE** menu button for the polarity-reversal supervision type. Or, press 2 or the **VALID CALL** menu button for Valid Call timer supervision. Or, press 3 or the **VALID + POLARITY** menu button for Valid Call timer with Polarity Reversal supervision. Then press 7 or **ACCEPT** to continue.
  
  - **Caller ID:** To enable or disable Caller ID, press 2 or the **CALLER ID** menu button. When the display shows **SET CALLER ID**, press 1 or **ON** to enable the feature, or press 2 or **OFF** to disable it. Then press 7 or **ACCEPT** to continue.
  
  - **Hybrid Balance:** To set the hybrid balance of the trunk to normal or short, press 3 or the **HYBRID BALANCE** menu button. When the display shows **HYBRID BALANCE OPTION**, press 1 or the **NORMAL** menu button, or press 2 or the **SHORT** menu button. Then press 7 or **ACCEPT** to continue.
  
  - **DTMF/Dial-Pulse Signaling:** To set the trunk for DTMF or dial-pulse signaling, press 4 or the **SIGNALING** menu button. When the display shows **ENTER SIGNALING OPTION**, press 1 or the **DTMF** menu button, or press 2 or the **PULSE** menu button. Then press 7 or **ACCEPT** to continue.
  
  - **Trunk Group Number:** To assign the trunk to a trunk group, press 5 or the **TRUNK GROUP** menu button. When the display shows **ENTER TRUNK GRP EXTENSION**, enter the trunk group number you want using one of the methods described on page 43.
  
  - **Busy Trunk:** To temporarily remove the trunk from service, press 6 or the **BUSY TRUNK** menu button. When the display shows **MARK TRUNK BUSY**, press 1 or the **ON** menu button, or press 2 or the **OFF** menu button. Then press 7 or **ACCEPT** to continue.

- To exit, press 7 or **ACCEPT** to move backward through the prompts until you reach the **ENTER DATABASE OPTION** prompt.
5. **To Program A Trunk Group:** Press 2 or the TRUNK GROUP menu button. Then program the trunk group, as follows:

- The display shows ENTER TRUNK GRP EXTENSION. Enter the extension number of the trunk group to be programmed using one of the methods described on page 43.
- The display shows ENTER TRUNK GROUP OPTION. Select one of the following options. To view the last three options on a display endpoint, press the arrow (<< or >>) menu button at the bottom of the display.
  - **Answer Access For Day Or Night Mode:** To program the endpoint(s) that will have answer access for this trunk group for day mode, press 1 or the ANS-ACCESS DAY menu button. To program for night mode, press 2 or the ANS-ACCESS NIGHT menu button. When the display shows ENTER ENDPOINT EXTENSION, perform the procedure described below.
    
    a.) Enter an endpoint extension using one of the methods described on page 43.

    b.) The display shows the extension number and a Yes/No prompt. To include the endpoint in the list, press 1 or the YES menu button, followed by #. Or, to exclude the endpoint from the list, press 2 or the NO menu button, followed by #.

    c.) When the display returns to ENTER ENDPOINT EXTENSION, select another extension number. Or, if finished programming the endpoint list, press 7 or ACCEPT to continue.
  
  - **Ring-In For Day Or Night Mode:** To program the endpoint that will have ring in for this trunk group for day mode, press 3 or the RING-IN DAY menu button. To program for night mode, press 4 or the RING-IN NIGHT menu button. When the display shows ENTER ENDPOINT EXTENSION, enter endpoints into the list one at a time as described above for Answer Access.
  
  - **Toll Restriction:** To program the toll restrictions for the trunk group, press 5 or the TOLL RESTRICTION menu button. When the display shows TOLL RESTRICTION OPTION, press the numbers that correspond to the toll restrictions you want to program, as follows:

    a.) **COS for day or night mode:** To program the toll restriction class of service for day mode, press 1 or the COS DAY menu button. To program class of service for night mode, press 2 or the COS NIGHT menu button. The display shows SET DAY (or NIGHT) COS XX. If this is not the COS you want to program, scroll to the correct COS by pressing the high end of the Volume button or the NEXT menu button to scroll forward, or press the low end of the Volume button or the PREVIOUS menu button to scroll backward. The default COS numbers are listed below. For definitions, see “Class of Service” on page 143. COS 01 is not used for trunks. When the correct COS is displayed, press 1 or the ON menu button to enable the toll restriction. Or, press 2 or the OFF menu button to disable it. Repeat this step to program additional COS toll restrictions, and then press 7 or ACCEPT to continue.
For U.S. COS options, use the following values:

**COS 02 – Deny Area/Office**

**COS 03 – Deny Operator**

**COS 04 – Deny Toll Access**

**COS 05 – Deny International**

**COS 06 – Deny Equal Access**

**COS 07 – Deny Local Calls**

**COS 08 – Denied Numbers**

**COS 09 – Allowed Numbers**

For European COS options, use the following values:

**COS 01 – ARS Only**

**COS 02 – Deny Operator**

**COS 03 – Deny Toll Access**

**COS 04 – Deny International**

**COS 05 – Deny Local Calls**

**COS 06 – Denied Numbers**

**COS 07 – Allowed Numbers**

b.) *Exempt from ARS*: To program the Exempt from ARS option, press 3 or the **EXEMPT FROM ARS** button. When the display shows **EXEMPT FROM ARS**, press 1 or the **YES** menu button to enable the option. Or, press 2 or the **NO** menu button to disable it. Then press 7 or **ACCEPT** to continue.

c.) *Subject to toll restriction*: To program the trunk group Subject to Toll Restriction option, press 4 or the **SUBJECT TO TOLL** menu button. When the display shows **SUBJECT TO TOLL**, press 1 or the **YES** menu button to enable the option. Or, press 2 or the **NO** menu button to disable it. Then press 7 or **ACCEPT** to continue.

– **Trunk List**: To program the list of trunks for the trunk group, press 5 or the **TRUNK LIST** menu button. When the display shows **ENTER TRUNK EXTENSION**, enter trunks into the list as follows:

a.) Enter a trunk extension using one of the methods described on page 43.

b.) The display shows the extension number and a Yes/No prompt. Press 1 or the **YES** menu button, followed by 7, to include the trunk in the list. Or, press 2 or the **NO** menu button, followed by 7, to exclude the trunk from the list.

c.) When the display returns to **ENTER TRUNK EXTENSION**, select another extension number. Or, if finished programming the trunk list, press 7 or **ACCEPT** to continue.

- To exit, press 7 or **ACCEPT** to move backward through the prompts until you reach the **ENTER DATABASE OPTION** prompt.
6. When finished with all programming, press ** while the ENTER DATABASE OPTION prompt is displayed. This ends the programming session.

7. If a system reset is required, the display shows ENTER SYS RESET OPTION. Do one of the following:
   - **Delayed Reset:** Press ** or the DELAYED menu button to delay the reset. The display shows DELAYED RESET SCHEDULED. The system will be reset at the pre-programmed time.
   - **Immediate Reset:** Press ** or the IMMEDIATE menu button to reset the system now.

**NOTICE**

Possible interruption of calls in progress. When the system is Reset, all calls in progress are dropped immediately.
Figure 3. Trunk Database Programming Flowchart

- Enter Program Database feature code 9932 [9132]

Press 3 or TRUNK menu button

Press 1 or INDIVIDUAL TRUNK menu button

Type the trunk extension number

Press 1 or INDIVIDUAL TRUNK menu button

Press 2 or TRUNK GROUP menu button

Enter the trunk group number

AT ANY LEVEL:
Press * or the Speaker button to cancel any unsaved changes and back up to exit programming.

Press # or ACCEPT to save changes and back up one display level.

INFORMATION IN THIS CHART IS SHOWN AS FOLLOWS:

ENDPOINT DISPLAYS

YOUR INPUT
Figure 3. Trunk Database Programming Flowchart (Continued)

CONTINUED FROM PREVIOUS PAGE

Enter the number of the trunk group to be programmed

TRUNK GROUP OPTION

Press 1 or ANS ACCESS DAY menu button
Press 2 or ANS ACCESS NIGHT menu button
Press 3 or RING-IN DAY menu button
Press 4 or RING-IN NIGHT menu button

ENTER ENDPOINT EXTENSION

Type the extension number

(Extension number) YES/NO

Press 1 or YES menu button
Press 2 or NO menu button

TOLL RESTRICTION OPTION

Press 5 or TOLL RESTRICTION menu button

SET DAY/NIGHT COS XX

If necessary, select COS number to program by scrolling

Press 1 or ON menu button
Press 2 or OFF menu button

EXEMPT FROM ARS

Press 1 or YES menu button
Press 2 or NO menu button

SUBJECT TO TOLL

Press 1 or YES menu button
Press 2 or NO menu button

ENTER TRUNK EXTENSION

Type the trunk number

(Extension number) YES/NO

Press 1 or YES menu button
Press 2 or NO menu button

ENTER ENDPOINT EXTENSION

Type the extension number

(Extension number) YES/NO

Press 1 or YES menu button
Press 2 or NO menu button

ENTER ENDPOINT EXTENSION

Type the extension number

(Extension number) YES/NO

Press 1 or YES menu button
Press 2 or NO menu button

ENTER ENDPOINT EXTENSION

Type the extension number

(Extension number) YES/NO

Press 1 or YES menu button
Press 2 or NO menu button

ENTER ENDPOINT EXTENSION

Type the extension number

(Extension number) YES/NO

Press 1 or YES menu button
Press 2 or NO menu button

ENTER TRUNK EXTENSION

Type the trunk number

(Extension number) YES/NO

Press 1 or YES menu button
Press 2 or NO menu button
Programming System Speed Dial

System Speed Dial numbers and names can be programmed by the installer or at any adminis-
trator endpoint.

Speed Dial Numbers

System Speed Dial numbers can contain up to 48 digits each and can include hookflashes and/
or pauses for dialing a series of numbers. For example, the Speed Dial number can contain an
SCC local number, a pause, an access code, and the phone number. To use PBX trunks in
Speed Dial numbers, enter the other system’s trunk access codes followed by a pause. The
lengths of the hookflash and the pause are determined by the programmable CO Hookflash and
Pause Digit timers.

To program System Speed Dial numbers, use one of the following methods:

- Use the dialpad to manually dial the number. To backspace, press MUTE.
- Press REDIAL to enter the last number dialed or saved at the endpoint (up to 48 digits).
- Press one of the Speed Dial buttons on the Inter-Tel endpoint to enter the outside phone
  number (up to 16 digits) programmed under that button.

Speed Dial Names

Speed dial names can contain up to 16 characters.

Dialpad buttons are used to enter letters, numbers, and punctuation in English/Spanish, or Jap-
anese (Katakana). When programming Speed Dial names, you can press ALPHA MODE/USE NUMERIC
MODE menu button to switch between alphanumeric and numeric modes.

To program Speed Dial names:

- In numeric mode, the dialpad buttons are used to enter numbers 0–9, the pound (#) button is used for entering a hyphen (-), and the asterisk (*) button is used for entering a colon (:). For example, 1 00 would enter “1:00” in numeric mode.
- In alphanumeric mode, dialpad buttons are used to enter letters, numbers, and punctuation. The number of times a button is pressed determines which character is entered. For example, 533266 would enter “JEAN.” When adjoining characters are located under the same button, press FWD to advance to the next character. For example, 66777 66 would enter “NORM.”

If you make a mistake, use the MUTE button to backspace. Or, press CANCEL at any
time to cancel the operation (the name and number in memory remains unchanged and the dis-
play shows FEATURE CANCELED). Once the feature is cancelled, start over.
Table 9 shows the dialpad equivalents for English/Spanish and for Japanese language characters. Japanese characters are included in your standard software license and can be programmed as the Primary or Secondary Language.

**Table 9. Alphanumeric Equivalents for Dialpad Programming**

<table>
<thead>
<tr>
<th>Number of Times Button is Pressed</th>
<th>Button</th>
<th>ENGLISH/SPANISH CHARACTERS</th>
<th>KATAKANA CHARACTERS*</th>
</tr>
</thead>
<tbody>
<tr>
<td>1</td>
<td>-</td>
<td>A</td>
<td>a</td>
</tr>
<tr>
<td></td>
<td>&amp;</td>
<td>I</td>
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<td>2</td>
<td>A</td>
<td>KA</td>
<td>KA</td>
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<tr>
<td></td>
<td>B</td>
<td>KI</td>
<td>KI</td>
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<td>C</td>
<td>KU</td>
<td>KU</td>
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<td></td>
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<td>KE</td>
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<td></td>
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<td>KO</td>
<td>KO</td>
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<td>3</td>
<td>D</td>
<td>SA</td>
<td>SA</td>
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<td>TO</td>
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<td>5</td>
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<td>9</td>
<td>W</td>
<td>RA</td>
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<td></td>
<td>X</td>
<td>RI</td>
<td>RI</td>
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<td>Y</td>
<td>RU</td>
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<td>Z</td>
<td>RE</td>
<td>RE</td>
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<td></td>
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<td>RO</td>
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<td>yo</td>
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<td>WO</td>
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<td>ba</td>
</tr>
<tr>
<td></td>
<td></td>
<td>long</td>
<td>long</td>
</tr>
</tbody>
</table>

* Japanese characters are available only if the Multilingual feature is enabled and Japanese is installed as the secondary language.

**To program or change System Speed Dial numbers and names:**

1. While on-hook, enter the Program System Speed Dial feature code (9801). Display endpoints show PROGRAM SYS SPD # (000–999).

2. Enter the Speed Dial location code (000–999). The display shows ENTER NAME: <current name>. If you do not continue to the next step, the display times out after the Long Inter-digit timer expires and you must start over.

3. Perform the following actions, as needed:

   a. **To change or program the name:**

      Using your dialpad as described on page 50, type the name for the Speed Dial number you want, and then press # or ACCEPT. See Table 9, above.

      *If you make a mistake*, use the MUTE button to back up and erase existing characters or the CLEAR button to erase the whole entry.

   b. **To erase the current name and leave it blank:**

      Press CLEAR or press MUTE repeatedly until the name is erased, and then press # or ACCEPT.

   c. **To leave the name the same:**

      Press # or ACCEPT.
4. When the display shows ENTER NUMBER, enter the number, up to 48 digits, to be stored using the dialpad, one of the Speed Dial buttons, or the REDIAL button. If necessary, use the MUTE button to back up and erase existing numbers. (Display endpoints show the number as it is entered.) To include *, #, pause, or hookflash in the number, press the Special button (Redial) once for an asterisk (*), twice for a pound (#), three times for a hookflash (F), or four times for a pause (P). Executive Display, Professional Display or Model 8560 endpoint users can press the #, PLAN, FLASH, or PAUSE menu buttons to enter special characters. You cannot exceed 48 digits.

5. Press or ACCEPT to save the number. You hear a single progress tone when the number is accepted. Your display shows SYS SPD BIN #XXX UPDATED.

To erase System Speed Dial names or numbers:

1. While on-hook, enter the Program System Speed Dial feature code (9801). Display endpoints show PROGRAM SYS SPD # (000–999).

2. Enter the Speed Dial location code (000–999). The display shows ENTER NAME: <current name>.

3. You may erase the name and/or number:

To erase the name: Press MUTE repeatedly until the name is erased or press CLEAR.

To erase the number: Press or ACCEPT to display the ENTER NUMBER prompt, then press MUTE repeatedly until the number is erased or press CLEAR.

4. Press or ACCEPT to exit. You hear a single progress tone when accepted. Your display shows SYS SPD BIN #XXX UPDATED.

Programming and Using Mini-DSS and DSS/BLF Buttons

Direct Station Select (DSS) buttons, located on a Mini-DSS, a DSS/BLF Unit or on an Inter-Tel endpoint, provide one-button access to extension numbers. If the button is equipped with a lamp, you can visually monitor the status of the associated extension. Mini-DSS buttons are not available on single line endpoints.

You can program a DSS button for an off-node device, but not for “wildcard” off-node extensions.

Together, the lamps in the buttons create a busy lamp field that indicates the status of the resource (e.g., endpoint, trunk, hunt group, or feature) assigned to the buttons. The table shown below explains what Mini-DSS lamps indicate about the associated resource.

Table 10. Mini-DSS Lamp Meanings

<table>
<thead>
<tr>
<th>If the Lamp is . . .</th>
<th>Then . . .</th>
</tr>
</thead>
<tbody>
<tr>
<td>Solidly Lit</td>
<td>The associated resource is busy or all endpoints in the associated hunt group are unavailable in Do-Not-Disturb, busy, or with hunt group remove enabled. Or, the associated feature is enabled.</td>
</tr>
<tr>
<td>Slow Flash</td>
<td>The associated resource is in Do-Not-Disturb.</td>
</tr>
<tr>
<td>Fast Flash</td>
<td>The associated resource has a call ringing in or a call is camped on to the hunt group. This option is programmable. If DSS ring flash is disabled, a call ringing in to an endpoint shows a solidly lit lamp.</td>
</tr>
<tr>
<td>Continuous Flutter</td>
<td>The associated resource is causing an ENDPOINT OFF-HOOK system alarm.</td>
</tr>
</tbody>
</table>
In DB Programming, Mini-DSS and DSS/BLF buttons may also be programmed for direct access to other resources such as paging zones, or they can be assigned feature codes. See “Endpoint Feature Codes” on page 316. Busy lamp indication is provided to show the idle or busy status of the resource or feature.

Programming a DSS button as a resource does not always provide a transfer capability. For example, a DSS button programmed for paging will not transfer a party to the paging system.

Programming DSS buttons:
1. With the handset in the cradle, dial 3 1 7.
2. Press the DSS button and then dial the extension number you want.

Once programmed, you can use your DSS button as follows:
- You can place an intercom call to the extension that the button is programmed for by pressing the button once.
- You can transfer a call to the extension that the button is programmed for by pressing the button once and hanging up. For transferred calls that are forwarded to Voice Mail you will hear repeating double tones, and the display shows DEST FORWARDED TO VOICE MAIL.
- If the button has a lamp, you can visually monitor the status of the extension.

Placing an intercom call using DSS buttons:
1. To use the handset: Lift the handset. The display shows ENTER EXTENSION NUMBER.
2. If you want to place a private (nonhandsfree) call to an Inter-Tel endpoint, press the pound (♯) button. The display shows RING EXTENSION NUMBER.
3. Press the button for the number you want to ring.

If calling an Inter-Tel endpoint with handsfree answering enabled, you are immediately connected.
If placing a private call, calling a hunt group, calling a single line endpoint, or calling an Inter-Tel endpoint with the Handsfree feature disabled, you hear repeating double tones until the call is answered.
If the called OHVA-enabled endpoint is busy and your endpoint is enabled for immediate OHVA transmit, do not hang up when calling a busy endpoint. Press the Mini-DSS button again. If the endpoint’s secondary voice path is available, you are immediately connected for an off-hook voice announce call and may speak.

Transferring calls to stations:
1. While on a call, press the Mini-DSS button you want. The call is placed on transfer hold.
   a. Transfer to a station that is forwarded to Voice Mail: You hear repeating double tones and the display shows DEST FORWARDED TO VOICE MAIL.
   b. Transfer to Voice Mail: The system waits for you to enter the mailbox number. The display shows ENTER MAILBOX. Type the mailbox number you want. If you do not enter a mailbox number before you hang up, the caller will be connected to the Voice Mail unit and must enter the mailbox number after listening to the introductory prompts.
2. If necessary, wait for an answer and announce the call. If you are calling a device that does not accept handsfree calls (such as Automated Attendant location), the display shows HANG UP TO FINISH TRANSFER.

3. You have the following options:

To complete the transfer: Hang up or press another Call button. The display shows CALL TRANSFERRED TO <username>.

To transfer the call to hold: Press the Hold button or press \ and enter the Individual Hold feature code (336). The call will not ring or send call waiting signals until the Hold timer expires.

To try another station: Press another Mini-DSS button.

To return to the caller: Press the lit Call button, trunk button, or \ button.

Reverse transferring (picking up a call ringing or holding at another extension):

EITHER, enter the Reverse Transfer feature code (4) and then press the DSS/BLF button of the station where the call is ringing.

OR, press the Mini-DSS button you want, and then press \ and enter the Reverse Transfer feature code (4).

Managing System and Network Alarms

To allow one administrator to monitor multiple nodes, the system provides both system alarms and network-wide alarms.

System Alarms

The system’s Alarm Reporting feature detects equipment failures. If there is a system failure that affects service, a major alarm is displayed at all affected endpoints.

When a minor equipment failure occurs, a minor alarm is generated and appears on the primary attendant’s display and, if enabled, at administrator endpoints. If enabled in the Message Print programming, the alarm also appears in the error/message report.

Alarms 1–99 indicate problems that can be corrected without calling service personnel. Alarms 100–199 (telephone system) and 200–299 (voice processing system) require attention from service personnel.

When a Voice Mail alarm is in effect, the telephone system may be functioning properly, but the voice processing system may be inoperative.

Network Alarms

Each node has two flags which determine whether the node broadcasts or receives network-wide alarms.

- The Send Network Alarms flag determines whether a node will broadcast alarms that occur on that node to the rest of the network.

- The Receive Network Alarms flag determines whether the node will receive and display alarms sent by other nodes in the network.

To differentiate between network-wide and local alarms, network-wide alarms appear on administrator endpoints preceded by NET ALARM, and local system alarms are preceded with SYS ALARM. Network-wide alarms will override system alarms on an administrator endpoint display.
On remote nodes, network-wide alarms will indicate the name of the node on which the alarm occurred. The node name is obtained from the username.

**To respond to a minor alarm from any alarm display endpoint:**

1. When a minor alarm indication appears, write down the alarm information.
2. While on-hook, clear the alarm by entering the Clear System Alarm feature code, 9850, or the Clear Network Alarm feature code, 9851. The display shows SYSTEM (or NETWORK) ALARM CLEARED.
3. Take the appropriate action as indicated in Table 49 on page 307.

**To respond to a major alarm from any alarm display endpoint:**

Take the action defined in Table 11.

<table>
<thead>
<tr>
<th>Alarm Text</th>
<th>Description and Action Required</th>
</tr>
</thead>
<tbody>
<tr>
<td>#100–199; #200–224</td>
<td>These alarms indicate that an equipment failure has occurred that requires the attention of service personnel.</td>
</tr>
<tr>
<td>SYS ALARM #NNN</td>
<td>Contact service personnel.</td>
</tr>
<tr>
<td>CALL TECHNICIAN</td>
<td>Be prepared to provide all alarm information, as well as what was taking place when the alarm occurred.</td>
</tr>
</tbody>
</table>

A major alarm message, MAJOR ALARM, appears on all display endpoints in the event of a system-wide failure. If the major alarm appears on a group of endpoints, the associated Digital Endpoint Module (DEM-16) may have failed. The warning might also appear on a single endpoint if the endpoint or its cabling are defective. Because they indicate that all or part of the system is inoperative, major alarms require immediate attention from service personnel.

**Clearing Alarms**

An administrator can clear a network-wide alarm on the local node only or on every node in the network. To clear an alarm, you can use either the administrator feature codes or the menu buttons on your endpoint.

**To use the feature codes to clear alarms:**

Enter the appropriate feature code to clear the alarm. The clear alarms feature codes are as follows:

- **Clear Network Alarm (9851):** This feature code clears network-wide alarms on every node in the network, but will not affect system alarms. The Clear Network Alarm feature code may be entered on any node in the network, but the Send Network Alarms flag must be set for you to clear alarms on other nodes in the network.

- **Clear System Alarm (9850):** This feature code clears all local and network-wide system alarm displays on your node.

After you enter the feature code, the display shows SYSTEM (or NETWORK) ALARM CLEARED, and the highest priority alarm is cleared. If another alarm is active, the display shows that alarm. Enter the feature code again to clear the next alarm in the queue (only one alarm is cleared at a time). When all alarms are cleared next highest priority alarm is displayed.
To use menu buttons to clear alarms, do one of the following:

- Press CLEAR ALARM to clear the currently displayed alarm. The display shows SYSTEM (or NETWORK) ALARM CLEARED, and the next highest priority alarm is displayed. If there are no more alarms, the display shows the idle menu.

  or

- Press CLEAR ALL ALARMS to clear all alarms. The display shows ALL ALARMS CLEARED, and the idle menu is shown.

You can press IDLE MENU at any time to view the idle menu instead of the clear alarm menu.

Alarm Queue

The Alarm Queue feature prioritizes system and network alarms based on severity and allows system administrators to view and handle critical alarms before addressing minor alarms. You can then clear the individual alarm, or clear all the alarms in the queue, up to 30. When clearing alarms individually, the alarms are displayed in order of severity. When an extension has dialed emergency services, the Emergency Alarm is the only priority 1 alarm. Other prioritized alarms have a 2, 3, or 4 priority, based on the severity of the alarm.

Not all alarms have a number 1, 2, 3, or 4 priority. The numbered priority scheme is limited to alarms that can cause a major or minor system reset. Those alarms in the Alarm Queue that have a numbered priority (1-4) are displayed before the alarms that do not receive a numbered priority. Alarms that do not have a numbered priority are prioritized in the queue by date and time.

When alarms are generated:

- The highest priority alarm is placed in the front of the queue, regardless of when lower priority alarms are generated. For example, if A114 (priority 3) and A116 (priority 4) are currently in the queue, but A119 (priority 2) is generated, A119 is placed first in the queue.

- Alarms with the same priority level are placed in the queue based on the time the alarm was generated. For example, if A010 (priority 3) is generated at 10:30 AM, and A012 (priority 3) is generated at 10:32 AM, A012 is placed in the queue after A010.

- If the queue contains 30 alarms, the oldest, lowest priority alarm is overwritten with the new alarm. For example, if the queue currently holds 30 alarms, 20 of which are priority 4, and a priority 3 alarm is generated, the oldest priority 4 alarm is overwritten.

- Repetitive alarms such as A125, are placed in the queue only once. If the alarm is regenerated, the alarm that is currently in the queue is overwritten with the new alarm data (if applicable) and time. For example, A125 is overwritten each time it is regenerated, which is every five minutes. This prevents the queue from being filled with duplicate alarms.
The queue holds up to 30 alarms, with priority, 1 (critical) to 4 (low), as shown in the following lists.

<table>
<thead>
<tr>
<th>Alarm #</th>
<th>Priority</th>
<th>Alarm #</th>
<th>Priority</th>
</tr>
</thead>
<tbody>
<tr>
<td>A010</td>
<td>Low</td>
<td>A110</td>
<td>Med</td>
</tr>
<tr>
<td>A011</td>
<td>Critical</td>
<td>A111</td>
<td>Med</td>
</tr>
<tr>
<td>A012</td>
<td>Low</td>
<td>A112</td>
<td>Med</td>
</tr>
<tr>
<td>A013</td>
<td>Low</td>
<td>A114</td>
<td>Med</td>
</tr>
<tr>
<td>A014</td>
<td>Low</td>
<td>A115</td>
<td>Med</td>
</tr>
<tr>
<td>A015</td>
<td>Low</td>
<td>A116</td>
<td>Low</td>
</tr>
<tr>
<td>A016</td>
<td>Low</td>
<td>A117</td>
<td>Low</td>
</tr>
<tr>
<td>A017</td>
<td>Low</td>
<td>A118</td>
<td>Low</td>
</tr>
<tr>
<td>A018</td>
<td>Low</td>
<td>A119</td>
<td>High</td>
</tr>
<tr>
<td>A020</td>
<td>Low</td>
<td>A120</td>
<td>Med</td>
</tr>
<tr>
<td>A021</td>
<td>Low</td>
<td>A121</td>
<td>High</td>
</tr>
<tr>
<td>A022</td>
<td>Low</td>
<td>A122</td>
<td>High</td>
</tr>
<tr>
<td>A023</td>
<td>Low</td>
<td>A123</td>
<td>High</td>
</tr>
<tr>
<td>A024</td>
<td>Low</td>
<td>A124</td>
<td>Med</td>
</tr>
<tr>
<td>A025</td>
<td>Low</td>
<td>A125</td>
<td>High</td>
</tr>
<tr>
<td>A026</td>
<td>Low</td>
<td>A126</td>
<td>Med</td>
</tr>
<tr>
<td>A031</td>
<td>Low</td>
<td>A127</td>
<td>Med</td>
</tr>
<tr>
<td>A032</td>
<td>Low</td>
<td>A128</td>
<td>Low</td>
</tr>
<tr>
<td>A100</td>
<td>Low</td>
<td>A134</td>
<td>High</td>
</tr>
<tr>
<td>A101</td>
<td>Low</td>
<td>A135</td>
<td>High</td>
</tr>
<tr>
<td>A102</td>
<td>Low</td>
<td>A137</td>
<td>High</td>
</tr>
<tr>
<td>A103</td>
<td>Low</td>
<td>A138</td>
<td>High</td>
</tr>
<tr>
<td>A104</td>
<td>Med</td>
<td>A200</td>
<td>Low</td>
</tr>
<tr>
<td>A105</td>
<td>Low</td>
<td>A201</td>
<td>Low</td>
</tr>
<tr>
<td>A106</td>
<td>Low</td>
<td>A202</td>
<td>Low</td>
</tr>
<tr>
<td>A107</td>
<td>Low</td>
<td>A203</td>
<td>High</td>
</tr>
<tr>
<td>A108</td>
<td>Low</td>
<td>A204</td>
<td>High</td>
</tr>
<tr>
<td>A109</td>
<td>Med</td>
<td></td>
<td></td>
</tr>
</tbody>
</table>
Networked Inter-Tel 5000 and Axxess systems containing nodes running Inter-Tel Axxess v8.1 or earlier require special attention because those nodes do not have the Alarm Queue feature.

In networks that include these early-version Axxess systems, a broadcast message is sent only when the most recent network alarm is cleared. When the message is sent, that network alarm is cleared on all nodes in the system. After the alarm is cleared, Axxess v8.1 or earlier nodes will not show any other alarms because they do not have a queue to store alarms.

Axxess nodes that are running v8.2 and later, however, may still have additional alarms in the queue. When this happens, each v8.2-and-later node must manually clear the other network alarms.

For example, imagine a three-node network with Node 1 running Axxess v8.0, Node 2 running Axxess v9.0, and Node 3 running Inter-Tel 5000 v1.1 or later. The following sequence of events could possibly occur:

- Node 1 generates two network alarms, one immediately after the other. However, Node 1 shows only the second alarm because it has no ability to queue alarms. Nodes 2 and 3 show both network alarms.
- Node 1 clears the second alarm—the only one it can recognize—and a broadcast message is sent to the other nodes. Node 1 now shows no alarms, but Nodes 2 and 3 still show the first alarm.
- Node 2 then clears the first alarm, but a broadcast message is NOT sent.
- Now, Nodes 1 and 2 show no alarms, but Node 3 still has one in its alarm queue. This alarm will stay in the queue until Node 3 clears it or until the alarm is overwritten as a repetitive alarm.

### Enabling/Disabling the Internal Modem

The Inter-Tel 5000 internal modem performs multiple functions. When enabled, the modem is used for DB Programming, Message Print, SMDR, and for uploading software licenses.

If disabled, the modem and extension are placed into DND. In DND, the modem rejects all calls and does not allow camp ons. Any outside calls that attempt to ring a disabled modem are automatically transferred to the primary attendant.

**To enable/disable a modem:**

1. While on-hook, enter the Modem Enable feature code, 9866, or the Modem Disable feature code, 9867. The display shows ENABLE (DISABLE) MODEM EXT #.
2. Enter the extension of the modem to enable/disable it. The display shows MODEM ENABLED, or DISABLED.

   If an invalid modem extension is entered, the display shows INVALID MODEM EXTENSION.
Resetting the Internal Modem

The internal modem must be reset if it fails to answer an incoming call or is out of sync with an external modem.

**NOTICE**

**Possible data corruption.** Do not reset the modem if it is connected to a DB Programming session. If you do, the programming session is dropped, potentially corrupting the database. For this reason, Inter-Tel recommends that you assign the modem extension to a Speed Dial button. During a reset, any call connected to the modem is dropped.

To reset the internal modem:
1. Enter the Modem Reset feature code, 9869. The display shows **RESET MODEM EXT#**.
2. Enter the modem extension number. The display shows **MODEM RESET**.
   
   If an invalid modem extension is entered, the display shows **INVALID MODEM EXTENSION**.

**Using the Network Group Diagnostics Feature**

The Network Group Diagnostics feature allows an administrator to verify that Network Groups have been programmed correctly. When the Network Group Diagnostics feature code is entered, the platform initiates pings from each Inter-Tel IP device and determines if the other devices respond to the ping. If a device does not respond to the ping or if a firewall is detected, the system issues a Message Print message. For details, see “System Error/Message Print” on page 254.

**IMPORTANT**

Because this feature affects system performance, Inter-Tel recommends that you run the diagnostics program after normal business hours when the system is idle.

The Network Group Diagnostics feature works on Inter-Tel proprietary IP devices for the local node only. It does not work across nodes.

**To run the Network Group Diagnostics:**

1. While on-hook, enter the Diagnostics Mode feature code, 9900 [9100 in Europe], to turn on diagnostics mode. The display shows **DIAGNOSTICS ON**.
2. Enter the Network Group Diagnostics feature code, 9963 [9163]. The display shows **NET GRP CHECK (YES = 1 NO = 2)**.
3. When prompted, press **1** or **ACCEPT** to start the diagnostics feature. The display shows **NET GROUP DIAG IN PROGRESS**.

When the diagnostics are complete, the endpoint displays one of the following messages:

- **NET GROUP CHECK COMPLETED**: Indicates that all IP devices within the Network Groups are capable of communicating via P2P audio.
- **NET GROUP CHECK ERRORS FOUND**: Indicates that either some of the IP devices are offline or there are NATs/firewalls located between the devices. Check Message Print to determine which errors occurred.
Seizing a Device

For troubleshooting purposes, the Seize Device feature code allows an administrator to enable diagnostics mode and then seize a specific trunk or extension by entering the module, circuit, and device number.

The Seize Device is affected by trunk restriction. The administrator endpoint must have outgoing access permission for the trunk to seize it. Toll restriction is applied to any calls placed after the trunk is seized.

To seize a device:

1. While on-hook, enter the Diagnostics Mode feature code, 9900 [9100], to turn on diagnostics mode. The display shows DIAGNOSTICS ON.

2. Press the Special button ( ) and enter the Seize Device feature code, 9973 [9173]. The display shows ENTER MODULE.

3. Enter the module number where the device to be seized is located, while observing the following conditions:
   - If you enter a single digit, press after the digit to continue.
   - If you enter an invalid module number, the display shows INVALID MODULE NUMBER.
   - Do not enter the Unit number (0). The display shows ENTER PORT.

4. Enter the port (circuit) number of the device to be seized. If you enter a single digit, press after the digit to continue. Valid numbers depend on the type of module. If you enter an invalid module number, the display shows INVALID MODULE NUMBER.

   If there is only one device on the selected port, the selected device is seized as if the user had directly entered the extension or trunk number.

   If there is more than one device on the selected port, the display shows ENTER DEVICE. Enter the number of the device to be seized. Valid numbers depend on the type of devices installed. If you enter an invalid device number, the display shows INVALID DEVICE NUMBER. When the device has been selected, it is seized as if the user had directly entered the extension or trunk access number.

5. Enter the Diagnostics Mode feature code, 9900 [9100], to turn off diagnostics mode. The display shows DIAGNOSTICS OFF.
Freezing/Unfreezing the System History

When certain system failures occur, service personnel might request a “system history freeze.” This allows them access to the system database so that they can analyze system activity for the time period preceding the alarm.

To freeze or unfreeze system history:

1. While on-hook, enter the Diagnostics Mode feature code, 9900 [9100], to turn on diagnostics mode. The display shows DIAGNOSTICS ON.

2. Enter the Freeze System History feature code, 9993 [9193], or Unfreeze System History feature code, 9998 [9198]. The display shows HISTORY FROZEN (or UNFROZEN). If you attempt to change the state of the system history to its current status, the display shows HISTORY ALREADY FROZEN (or UNFROZEN).

3. Enter the Diagnostics Mode feature code, 9900 [9100], to turn off diagnostics mode. The display shows DIAGNOSTICS OFF.

Freezing/Unfreezing the Network History

Administrators can freeze or unfreeze the fault history for any node in the network. Service personnel can then pull the information from each node for troubleshooting purposes. Nodes are assigned to freeze zones by the database programmer to allow selective freezes. You must know which zone to freeze/unfreeze and then enter the freeze feature code followed by the zone number.

A node can be in more than one freeze zone. If you freeze a zone that contains a node that is already frozen, the display shows SOME NODES ALREADY FROZEN and the node remains frozen. If you unfreeze a zone with unfrozen nodes in it, the display shows SOME NODES ALREADY UNFROZEN. To determine the individual freeze status of the nodes, check the message print output.

To freeze or unfreeze network history:

1. While on-hook, enter the Diagnostics Mode feature code, 9900 [9100], to turn on diagnostics mode. The display shows DIAGNOSTICS ON.

2. Enter the Network Freeze Zone System Histories feature code, 9939 [9139], or Network Unfreeze Zone System Histories feature code, 9989 [9189]. The display shows SELECT FRZ ZONE (0–9) #.

3. Enter the number of the zone you want to freeze or unfreeze. The display shows FREEZE ZONE XX IS FROZEN (or UNFROZEN). If you attempt change the state of the system history to its current status, the display shows FREEZE ZONE XX ALREADY FROZEN (or UNFROZEN).

4. Enter the Diagnostics Mode feature code, 9900 [9100], to turn off diagnostics mode. The display shows DIAGNOSTICS OFF.
Voice Mail Administrator Procedures

If your endpoint has been programmed as the voice mail administrator, you can use special features that are not provided to other voice mailbox owners. The voice mail administrator mailbox has all the end-user mailbox features, plus the ability to do the following:

- Record a broadcast message
- Perform mailbox and group list maintenance
- Create and select custom audiotex recordings (voice mail company greetings, auto attendant recordings, call routing announcements, and hunt group overflow and announcement station recordings)
- Import fax documents
- Customize or reinstate voice mail prompts. If ASR is enabled for your mailbox, you have the option of issuing a spoken command or pressing a dialpad button. For ASR guidelines, see “Automatic Speech Recognition” on page 271.

To begin using the voice mail administrator’s mailbox you will need to first initialize the mailbox.

To initialize the voice mail administrator’s mailbox:

1. Dial the voice mail access number.
2. Say “Login” or press 3 to identify yourself as a subscriber.
3. Say or enter the voice mail administrator’s mailbox number.
4. Enter your default password (mailbox number) and then press #.
5. If you want a password, enter a new password using digits 0–9 (up to 12 digits), and then press # when you are done. The messaging system plays back your password. If you do not want to use a password, just say “Skip” or press #.

6. Say “Accept” or press # to accept the entry or say “Erase” or press 2 to erase and re-enter your password. The system prompts you to record your directory name.
7. After the tone, record your first and last names.
8. When prompted, do one of the following:
   - Say “Accept” or press # again to accept the name.
   - Say “Replay” or press 1 to replay the name you just recorded.
   - Say “Append” or press 2 to add to your name.
   - Say “Re-record” or press 3 to erase and re-record your name.

Once you have initialized your mailbox, you can access it and the voice mail administrator’s menu as indicated below.
To access the voice mail administrator’s mailbox and menu:
1. Dial the voice mail access number.
2. Say “Login” or press # to identify yourself as a subscriber.
3. Say or enter the voice mail administrator’s mailbox number.
4. Enter the password and then press #.
5. Say “Nine” or press 9 to reach the voice mail administrator menu. No prompt is given for entering the 9.

Recording a Broadcast Message
You have the ability to make a single recorded message and send it to all Standard and Receive-Only mailboxes. Extension IDs do not receive Broadcast Messages.

To record a broadcast message:
1. Access the voice mail administrator’s mailbox and menu.
2. Say “Broadcast Message” or press 1 to send a broadcast message.
3. Record the message.
   
   If you exceed the maximum allotted time for message length, a prompt will notify you and allow you to re-record the message or send it.
   
   If voice mail is full, a message will notify you that it cannot accept the message. Hang up and try again later.
4. When you have completed the message, press # for the following options:
   • Say “Replay” or press 1 to replay the message.
   • Say “Append” or press 2 to add to the message.
   • Say “Erase” or press 3 to erase and re-record the message.
   • Say “Accept” or press # to accept the message. You must use the Accept option for the message to be sent. When you hang up, the message is canceled.

Maintaining Mailboxes and Group Lists

Voice Mail Administrator Capabilities
Using the voice mail administrator’s mailbox, you may provide basic voice mailbox and group list maintenance. You do not need to know the password of a particular mailbox to perform maintenance on it.

Mailbox and group list maintenance includes the ability to:
• Record the name of a standard mailbox user, extension ID user, or Group List
• Record a primary and alternate mailbox greeting
• Set additional personal options for a mailbox

IMPORTANT
To provide system security, all mailboxes and extension IDs should have a password. To make the passwords difficult to guess, they should not match the mailbox number or consist of one digit repeated several times. This is especially important for the voice mail administrator’s mailbox, which allows programming access to other mailboxes.
Mailbox and Extension ID Personal Options

Voice Mail mailboxes provide personal options that allow the user to customize messaging functions. However, before personal mailbox options can be changed, the mailbox must be initialized by the user. For programming instructions to change personal options, refer to the applicable Inter-Tel Endpoint User Guide, which are listed under “Endpoint User Guides” on page 358.

Each subscriber’s mailbox has the following personal options:

- **Greetings**: A primary or alternate greeting is played when callers reach a user’s mailbox. Users can change a primary greeting at any time to reflect a change in status and/or record and enable an alternate greeting. If you and the user elect not to record a personal greeting, the default system greeting is used.

  The system has a database option called **Play Recording Instructions** that affects mailbox greetings. If this option is enabled, the “leave a message after the beep” system prompt plays following the primary or alternate greeting. If disabled, no prompt is played and the beep preceding recording occurs immediately after the primary or alternate greeting ends. The Play Recording Instructions flag is enabled by default.

- **Directory Name**: A recorded name is used to identify each mailbox in the directory and to verify the mailbox number when messages are addressed by non-subscribers and other subscribers. Although the name is recorded when a user first sets up the mailbox, you can change it at any time.

- **Password**: Passwords prevent unauthorized access to a mailbox. Although the mailbox user creates a personal password when they first set up the mailbox, you can change it at any time.

- **Message Envelope Options**: Each message is preceded by an “envelope” that can include the time and date the message was left, the source of the message, and additional information that is associated with the media type. Voice mail messages have a length envelope option, e-mail messages have a subject option, and faxes have a pages option. By default, all envelope options are enabled for each type of message.

- **Remote Messaging**: Remote Messaging is available for voice mail messages only and is enabled in DB Programming on a per-mailbox basis. Using Remote Messaging, the user can program voice mail to call the mailbox owner when a new message arrives. Using a “cascade”—a series of up to nine telephone numbers that Voice Mail calls until it successfully connects to a device such as a pager, answering machine, etc. If for some reason Voice Mail is not able to use remote notification, the mailbox receives a message stating that notification could not be completed. For more information about Remote Messaging options, refer to the applicable Inter-Tel endpoint user guide.

- **Fax Destination Number**: If an endpoint is programmed to receive incoming faxes, the user can program the fax destination number that will receive the faxes that are sent to the mailbox.

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**IMPORTANT**

To provide system security, all mailboxes and extension IDs should have a password. To make the passwords difficult to guess, they should not match the mailbox number or consist of one digit repeated several times. You should recommend to all users that they set a password.
• **Message Search Order**: This option allows the user to determine how to retrieve saved or new messages based on the date and time they were received. The message search order can be the earliest-received messages first (First In/First Out), or the latest-received messages first (Last In/First Out).

• **Transfer Method**: The Transfer Method personal option determines how the system will handle incoming calls that Enterprise Messaging (EM) may execute. For example, calls may be transferred to a mailbox or Extension ID by voice mail reply, a Call Routing Application, Automated Attendant, operator transfer, etc. Transfer Method options include Unannounced, Announce-Only, and Screened. For usage instructions, see “Call Screening Options” on page 268.

**Change Procedures for Mailbox Personal Options**

The voice mail administrator can change personal options for individual mailboxes or for the entire Voice Mail system, as described in the following procedures.

**To record or change a personal greeting:**

1. Access the voice mail administrator’s mailbox and menu. See page 63.

2. Say “Mailbox Maintenance” or press 2 to access the Mailbox/Group List Maintenance option.

3. Say or enter the mailbox or extension ID to be accessed. You hear the Personal Options Menu.

4. Say “Greeting” or press 1 to record/change the personal greeting. You have the following options:
   - Say “Primary” or press 1 to record and/or enable the primary greeting.
   - Say “Alternate” or press 2 to record and/or enable the alternate greeting.
   - Say “System” or press 3 to enable the system default mailbox greeting.

5. If you selected the system greeting option, continue to the next step.

6. If you selected the primary or alternate greeting option, record the greeting when prompted, then press 9.

7. If you already have a primary or alternate greeting, it is played when you select primary or alternate.

You can then do one of the following:
   - Say “Accept” or press 1 to accept the greeting.
   - Say “Replay” or press 1 to replay the greeting.
   - Say “Append” or press 2 to add to the greeting. Press 4 when you are done.
   - Say “Erase” or press 3 to erase and re-record the greeting.
   - Say “Cancel” or press 4 to exit without changing the greeting.

6. Hang up or press 9 to return to the previous menu.
To change the recorded directory name for a mailbox:

1. Access the voice mail administrator’s mailbox and menu. See page 63.
2. Say “Mailbox Maintenance” or press 2 to access the Mailbox/Group List Maintenance option.
3. Say or enter the mailbox or extension ID to be accessed. You hear the Personal Options Menu.
4. Say “Name” or press 2. The recorded name is played. You have the following options:
   • Say “Accept” or press 1 to accept the recorded name.
   • Say “Replay” or press 1 to replay the recorded name.
   • Say “Append” or press 2 to add to the name.
   • Say “Erase” or press 3 to erase and re-record the name.
5. Hang up or press 4 to return to the previous menu.

To change a password:

1. Access the voice mail administrator’s mailbox and menu. See page 63.
2. Say “Mailbox Maintenance” or press 2 to access the Mailbox/Group List Maintenance option.
3. Say or enter the mailbox or extension ID to be accessed. You hear the Personal Options Menu.
4. Say “Password” or press 5.
5. If you want a password, enter a new password using digits 0–9 (up to 12 digits) and press 7 when you are done. Voice mail plays back the password.
   
   If you do not want to use a password, say “Skip” or press 4. Voice mail announces that you have chosen not to have a password.
6. Say “Accept” or press 7 to accept the password as entered, or say “Re-enter” or press 3 to erase and re-enter the password.
7. Hang up or press 4 to return to the previous menu.

To change and save a Message Envelope:

1. Access the voice mail administrator’s mailbox and menu. See page 63.
2. Say “Mailbox Maintenance” or press 2 to access the Mailbox/Group List Maintenance option.
3. Say or enter the mailbox or extension ID to be accessed. You hear the Personal Options Menu.
4. Say “Envelope” or press 4 to select the Message Envelope Options Menu.
5. Select the media type you would like to change:
   • Say “Voice Mail” or press 1.
   • Say “E-mail” or press 2.
   • Say “Facsimile” or press 3.
6. You can then do any of the following:
   - Say “Time/Date” or press 1 to enable or disable the time/date option.
   - Say “Source” or press 2 to enable or disable the message source option.
   - For voice mail, say “Length” or press 3 to enable or disable the message length option.
   - For e-mail, say “Subject” or press 3 to enable or disable the subject option.
   - For faxes, say “Pages” or press 3 to enable or disable the pages option.
   - Say “All Options” or press 4 to enable all options and return to the Personal Options Menu.
   - Say “None” or press 5 to disable the entire envelope and return to the Personal Options Menu.

   **NOTE** This option is not available for fax messages.

7. Hang up or press 4 to return to the previous menu.

**To set up Remote Messaging, if enabled in DB Programming:**

1. Access the voice mail administrator’s mailbox and menu. See page 63.
2. Say “Mailbox Maintenance” or press 2 to access the Mailbox/Group List Maintenance option.
3. Say or enter the mailbox or extension ID to be accessed. You hear the Personal Options menu.
4. Say “Remote” or press 5. You have the following options:
   - Say “Primary” or press 1 to set up a primary cascade.
   - Say “Alternate” or press 2 to set up an alternate cascade.
5. Say “Level” or press 1 and say or enter the number (1–9) of the level you want to program. You then have the following options:
   - To set up or change an extension, outside, or pager number:
     - Say “Number” or press 2.
     - **EITHER**, say “Internal” or press 1 for an extension number.
       - OR, say “Outside” or press 2 for an outside number.
     - Say or enter the number.
   - To set up or change pager notification:
     - Say “Pager” or press 1.
     - Say “Personal” or press 2.
   - To enable or disable the number, say “Change” or press 1.
6. Say “Time” or press 2. Then say or enter the time you want the message notification to start and stop. Enter or say the times with two digits for the hour and two digits for the minutes (e.g., 0900 = 9:00). If entering the time in 12-hour format:
   - Say “AM” or press 1 for a.m.
   - Say “PM” or press 2 for p.m. For 24-hour notification, set the starting and ending times to be the same.

7. Say “Days” or press 3. Then select one of the following:
   - Say “Weekdays” or press 1 for Monday-Friday.
   - Say “Days” or press 2 for all days.
   - Say “Day” or press 3 to select individual days. You are prompted to say or press digits 1–7 which correspond to the days Sunday through Saturday.

8. Say “Category” or press 4. You have the following options:
   - Say “All Messages” or press 1 for all messages.
   - Say “Priority” or press 2 for priority messages only.

9. Hang up or press 5 to save the settings and exit.

To change the Transfer Method, if enabled in DB Programming:

1. Access the voice mail administrator’s mailbox and menu. See page 63.

2. Say “Mailbox Maintenance” or press 2 to access the Mailbox/Group List Maintenance option.

3. Say or enter the mailbox or extension ID to be accessed. You hear the Personal Options menu.


5. Say “Transfer” or press 5. The system tells you what the current transfer method is. You can then select the appropriate transfer method, as follows:
   - Say “Unannounced” or press 1 to select unannounced transfers.
   - Say “Screened” or press 2 to select screened transfers.
   - Say “Announce” or press 3 to select announce only transfers.
   - Say “Cancel” or press 4 to return to the Personal Options Menu without making changes.

6. Hang up or press 5 to return to the previous menu.
To change a fax destination number:
1. Access the voice mail administrator’s mailbox and menu. See page 63.
2. Say “Mailbox Maintenance” or press 2 to access the Mailbox/Group List Maintenance option.
3. Say or enter the mailbox or extension ID to be accessed. You hear the Personal Options menu.
5. Say “Number” or press 1.
6. Say or enter the number of the fax destination.
7. EITHER, say “Accept” or press # when you are done.
   OR, say “Re-enter” and press 3 to erase and re-enter the number.
8. Hang up or press # to return to the previous menu.

To change the message search order:
1. Access the voice mail administrator’s mailbox and menu. See page 63.
2. Say “Mailbox Maintenance” or press 2 to access the Mailbox/Group List Maintenance option.
3. Say or enter the mailbox or extension ID to be accessed. You hear the Personal Options menu.
5. Say “Order” or press 2 for the Message Search Order.
6. Select one of the following options:
   • Say “New” or press 1 to change the search order for new messages.
   • Say “Saved” or press 2 to change the search order for saved messages.
7. Your current message search order is played. Select one of the following options:
   • Say “First In” or press 1 to retrieve the earliest-received messages first (First In/First Out).
   • Say “Last In” or press 2 to retrieve the latest-received messages first (Last In/First Out).
   • Say “Cancel” or press 3 to return to the Personal Options Menu without making changes.
8. Hang up or press # to return to the previous menu.
Changing a Group List Name

A Group List is a collection of mailboxes that is programmed in DB Programming. Subscribers can use Group Lists to send messages to several mailboxes simultaneously. With the voice mail administrator mailbox, you can record a name for the Group Lists that have been programmed in the system. However, before you can record a Group List name, you need to know the Group List number assigned to the list. Group List names are used for verification purposes only.

To record a group list name:
1. Access the voice mail administrator’s mailbox and menu. See page 63.
2. Say “Mailbox Maintenance” or press 2 to access the Mailbox/Group List Maintenance option.
3. Say or enter the Group List number to be accessed. The current name is played.
4. After the tone, record the name for the selected list, and then press #. You have the following options:
   - Say “Accept” or press 2 to accept the name.
   - Say “Replay” or press 1 to replay the name you just recorded.
   - Say “Erase” or press 3 to erase and re-record the name.
5. Hang up or press # to return to the previous menu.

Importing Fax Documents

If your voice mail system has fax enabled, you can use the voice mail administrator’s mailbox to import fax documents. Imported documents can then be requested by callers who use a Call Routing Announcement application programmed to provide the Fax-On-Demand service.

To import a fax document:
1. Using a fax machine, after you obtain dial tone, dial the voice mail access number.
2. Access the voice mail administrator’s mailbox and menu. See page 63.
3. Say “import fax” or press 4 to select the Import Fax option. (If the system does not have a fax license, this option is not accepted.) The system may prompt you with one of the following messages.
   - All system fax ports are busy: Hang up and try again later.
   - The system already contains the maximum number of fax documents allowed: Until one or more fax documents are deleted from the system, there is no room for additional documents.
   - The system does not have enough disk space to import any more documents: Until one or more fax documents are deleted from the system, there is no room for additional documents.
   - The system currently is at or above the Maximum Fax Library Size: Until one or more fax documents are deleted from the system, there is no room for additional documents. By default, the Maximum Fax Library Size field is set to 0 in DB Programming.
4. When prompted, say or enter the fax document number. One of the following occurs when you enter the number:

If the document number does not already exist, the system asks you to verify the number. Say “Yes” or press 4 if the number is correct and go to the next step. Or, say “Re-enter” or press 3 if you want to re-enter the number.

If the document number already exists, the system asks if you want to replace it. Say “Yes” or press 4 if you want to replace the document and go to the next step. Or, say “Re-enter” or press 3 if you want to re-enter the number.

If you did not enter a valid number, you are prompted to enter a new number.

5. When prompted, press the Start button on the fax machine. One of the following may occur:

If the system cannot communicate with your fax machine, check that it is working properly and say “Continue” or press 4 to try again. Return to step 4.

If the system runs out of disk space while importing the document or if the document will cause the system to exceed the Maximum Fax Library Size, you cannot import this document until one or more fax documents have been deleted from the system to create disk space. Say “Cancel” or press 7 to return to step 4.

6. When the fax transmission is complete, the system tells you that the document was received. Say “Continue” or press 4 to continue importing documents, or hang up.

Creating Audiotex Recordings

Using voice mail, you can create and select up to 500 custom audiotex recordings including voice mail company greetings, auto attendant greetings, call routing announcements, and hunt group announcement/overflow recordings. Depending on how your system is programmed and how the recordings will be used, you may need to record an ASR and a non-ASR version when you customize a recording.

Each recording has a designated number. Numbers are then assigned to specific applications using the voice mail administrator’s mailbox or through DB Programming. Any recording number (001–500) can be assigned to any voice processing system application. By default, recording #001 is the default system voice mail main greeting, and recording #002 is the Automated Attendant main greeting. A single recording may be used for multiple applications. The default audiotex recordings can be changed, but if the voice processing system database is defaulted, the system recordings will be restored.

After making the required recordings, you must determine which recordings will play during the telephone system’s day and night modes of operation. Programming day and night modes for audiotex recordings can be done from either the voice mail administrator’s mailbox or through DB Programming.

When customizing audiotex recordings, you must enter a valid three-digit recording number, or a single digit followed by # for recordings 1–9. If a valid number has been entered and a previous recording exists, the voice processing system will play it. This includes the two standard default greetings. When no recording is associated with a recording number, you are prompted to create a recording.

Once the recording has been made, the new recording replaces any previous recording. The only recordings that can be retrieved are system default recordings #001 and #002. They can be retrieved by defaulting the voice processing system database. The maximum recording time for each custom audiotex recording is determined in DB Programming.
In DB Programming, link several greetings together for one application and/or use the special UCD/ACD hunt group characters that tell the caller how many calls are ahead and how long the wait will be. If you use the voice mail administrator’s mailbox to record or assign a recording to an application with a list of several recordings, your input will change only the first entry in the list. You cannot change the other entries using the voice mail administrator’s mailbox. The custom audiotex program planning sheet is designed to help you plan and record custom recordings. See “Creating Audiotex Recordings” on page 71.

To make or change a custom audiotex recording:

1. Access the voice mail administrator’s mailbox and menu. See page 63.
2. Say “Options” or press 3 for recording options.
3. Say “Custom” or press 1 to make a recording.
4. When prompted, say or type the recording number (001–500) you want.
5. If no recording exists, go to the next step.
If a standard (non-ASR) recording already exists, it is played. You have the following options:
   • Say “Standard” or press 1 to record a standard (non-ASR) greeting.
   • Say “Copy” or press 2 to copy the standard (non-ASR) greeting.
   • Say “Speech Recognition” or press 3 to record an ASR greeting.
6. When prompted, record the greeting or message, and then press 4. While you are recording, you can:
   • Press 2 to pause while recording. (Say “Continue” or press 7 to continue.)
   • Press 3 to erase and re-record your message.
If you exceed the maximum allotted time for recording length, a prompt will notify you and allow you to re-record or save the recording.
If the voice processing system PC disk is full, a prompt will notify you that it cannot accept the recording. Hang up and try again later.
7. When finished recording, select one of the following options:
   • Say “Replay” or press 1 to replay the message.
   • Say “Append” or press 2 to add to the message.
   • Say “Erase” or press 3 to erase and re-record the message.
   • Say “Accept” or press 4 to accept.
8. To create an ASR recording, choose one of the following options:
   • Say “Copy” or press 2 to copy the standard (non-ASR) greeting.
   • Say “Speech Recognition” or press 3 to record an ASR greeting.
9. When finished, select one of the following options:
   - Say “Replay” or press 1 to replay the message.
   - Say “Append” or press 2 to add to the message.
   - Say “Erase” or press 3 to erase and re-record the message.
   - Say “Accept” or press 4 to accept.

To assign custom audiotex recordings to specific application extension numbers:

1. Access the voice mail administrator’s mailbox and menu. See page 63.
2. Say “Options” or press 3 for recording options.
3. Say “Select” or press 2 to select a recording.
4. When prompted, say or enter the extension number of the application to be programmed. (For example, if users dial 2000 to reach voice mail, enter 2000 to assign the voice mail greeting.)
5. Select the mode for the recording:
   - Say “Day” or press 1 for day mode.
   - Say “Night” or press 2 for night mode.
   - Say “Cancel” or press 7 to return to the previous step.
6. Say or type the recording number (001–500) you want. If a recording already exists, it is played. You have the following options:
   - Say “Yes” or press 7 to accept the recording.
   - Say “No” or press 8 to go back to step 4.

   If no recording exists, the system informs you, but assigns the recording number to the application. For instructions on how to make a custom recording, see “Creating Audiotex Recordings” on page 71.
7. Hang up or press 4 to return to the previous menu.

**Recording Custom Voice Mail Prompts**

The voice mail prompts consist of two sets of prerecorded options and instructions. One set of prompts is for applications and mailboxes that have Automatic Speech Recognition (ASR) enabled. The other set is for applications and mailboxes that do not use ASR. The prompts that use ASR give the caller the option to issue a spoken command or press a dialpad digit to complete a task. The prompts that do not use ASR prompt the caller to press a dialpad digit only.

Most of the prerecorded voice mail prompts can be customized, using the voice mail administrator’s mailbox. For prompts that cannot be replaced, you will hear a message informing you of the status. To keep the system prompts consistent, when you customize the non-ASR version of a prompt, you should customize the ASR version also, if one exists.

If the Multilingual feature is enabled, you can change either the Primary or Secondary Language prompts. To change primary language prompts, the endpoint must be in the primary language mode, likewise for secondary language prompts.
To record custom system prompts (or restore system prompts):

1. Access the voice mail administrator’s mailbox and menu. See page 63.
2. Say “Replacement” or press 0 to select the System Prompt Replacement option.
3. Say or type the prompt number (001–907) you want.
   *If the number is not valid*, you hear an error message and must choose another prompt number.
4. If the prompt is available in standard format only, go to the next step.
   If the prompt is available in two formats, select one of the following:
   • Say “Standard” or press 1 to record a standard (non-ASR) prompt.
   • Say “Speech Recognition” or press 3 to record an ASR prompt.
5. The current prompt is played. You have the following options:
   • Say “Replay” or press 1 to replay the prompt.
   • Say “Replace” or press 3 to replace the system prompt with a custom prompt.
     Record the prompt and then press 4.
   • Say “System” or press 7 to reinstate the system prompt.
   • Say “Cancel” or press 11 to leave the prompt unchanged.
   *If the voice processing system PC disk is full*, a prompt will notify you that it cannot accept the recording. Hang up and try again later.
   *If the newly created prompt is to be available in two formats*, repeat step 5 to record the other format.
6. When finished recording, you have the following options:
   • Say “Replay” or press 1 to replay the recording.
   • Say “Append” or press 2 to add to the message.
   • Say “Erase” or press 3 to erase and re-record the message.
   • Say “Accept” or press 2 to accept.

**Administrator Programming Planning Sheets**

The program planning sheets found in the Reference section are designed to help you plan your site’s system and voice mail programming. See page 309.
Hardware and Endpoints

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Introduction

The Inter-Tel 5000 platform can be licensed and programmed to support multi-protocol, IP, SIP, digital, and single line endpoints, softphone applications, as well as other endpoints and performance-enhancing hardware. The platform supports desktop and mobile endpoints that operate inside the local area network (LAN) or can be moved to the public network—the Internet—without reprogramming the system or jeopardizing network security.

Types of hardware supported include basic “hard” IP endpoints used by workers in business office environments. Some endpoints are multi-protocol—that is, they operate in either IP or SIP mode. Some operate in IP mode only and others operate in SIP mode only. Other endpoints support wireless IP endpoints, the Inter-Tel Model 8602 IP softphone application, and the Model 8601 SoftPhone for Pocket PC, which operates in SIP mode only on a Personal Digital Assistant (PDA). Single-Line Adapter endpoints support standard single line endpoints.

For attendant capability, the system supports a 16-extension direct station select (Mini-DSS) unit or 50-button or 60-button DSS/Busy Lamp Field (BLF) Units. The Mini-DSS attaches to one of the Inter-Tel six-line display IP or digital endpoints, and the DSS/BLF Units attach to digital display endpoints. To enable IP communication through the public telephone network, the Inter-Tel 5000 platform also supports gateway trunk hardware.

NOTE

For an explanation of what is meant by “endpoint,” see “Endpoints and Phones” on page 2.

To install and configure IP and SIP endpoints requires knowledge of appropriate network settings and familiarity with the associated hardware and software. Installation and configuration information for all IP and SIP endpoints should be referred to an Inter-Tel-certified service technician. For further information, contact your authorized Inter-Tel equipment provider.

For additional information, see “System Configurations” on page 12.

Base Server IP and SIP Emergency Calls From a Remote Site

To locate and quickly respond to a caller needing help, emergency response services depend on accurate information from Caller ID or, in Europe, Calling Line ID (CLID) features. Because of this reliance on Caller ID and CLID data, emergency responders can be misdirected or delayed if the Caller ID or CLID information provided to a dispatcher is incorrect.

To help ensure that emergency services can respond quickly to IP or SIP endpoint calls from remote sites, the remote site should be equipped with a correctly configured MGCP or SIP gateway. If a designated Emergency Call number is dialed from a remote site equipped with a correctly configured gateway, responders can be dispatched to the site of the emergency. However, if the remote site is not equipped with a correctly configured gateway, the call will be identified, instead, as coming from the location where the system Base Server is located.

NOTICE

Emergency Call phone numbers include:

- 911, the default for U.S. systems.
- 999, the default for UK systems.
- If applicable, 112, an emergency number used widely in Europe outside of the UK.

Any emergency dial number sequence that is appropriate for the location. Ask your Inter-Tel equipment and service provider or certified technician for details.
Also, if power to an Inter-Tel 5000 system fails, IP and SIP endpoints will not operate at remote sites or at the main system location without appropriate gateway equipment and programming, as described above. All IP and SIP endpoint users should be alerted to this potentially hazardous situation.

**IP Endpoints**

To allow system flexibility and cost efficiency, both Inter-Tel proprietary and other industry-standard endpoints can be used with the Inter-Tel 5000 platform.

IP endpoints operating in Inter-Tel Protocol (ITP) and Session Initiation Protocol (SIP) modes allow users to communicate across the local area networks (LANs) and the public telephone network. Hard IP and SIP endpoints can be identified by the additional ports on the back, which are used for inserting LAN and power connections. IP and SIP endpoints operate like digital endpoints installed on the system except for the limitations listed under “IP Endpoint Limitations” on page 85.

**ITP and SIP Supported Features**

Users can access most features by entering a feature code using the endpoint’s dialpad buttons. Table 12 shows the station features (and default feature codes) that are supported for endpoints operating in ITP mode and SIP mode.

Support for some features varies by endpoint. For a complete list of supported features refer to the appropriate user guide. For a summary of user guides and part numbers, see Table 55 on page 358.

**Table 12. ITP and SIP Features and Feature Codes**

<table>
<thead>
<tr>
<th>Feature Name</th>
<th>Feature Code</th>
<th>ITP Mode</th>
<th>SIP Mode</th>
</tr>
</thead>
<tbody>
<tr>
<td>Account Code – Following Calls</td>
<td>391</td>
<td>✓</td>
<td>x</td>
</tr>
<tr>
<td>Account Code – Optional</td>
<td>390</td>
<td>✓</td>
<td>x</td>
</tr>
<tr>
<td>ACD Agent Log In</td>
<td>326</td>
<td>✓</td>
<td>x</td>
</tr>
<tr>
<td>ACD Agent Log Out</td>
<td>327</td>
<td>✓</td>
<td>x</td>
</tr>
<tr>
<td>ACD Agent Log In/Out</td>
<td>328</td>
<td>✓</td>
<td>x</td>
</tr>
<tr>
<td>ACD Agent Wrap-Up Terminate</td>
<td>329</td>
<td>✓</td>
<td>x</td>
</tr>
<tr>
<td>Agent Help Request</td>
<td>375</td>
<td>✓</td>
<td>x</td>
</tr>
<tr>
<td>Agent Help Reject</td>
<td>376</td>
<td>✓</td>
<td>x</td>
</tr>
<tr>
<td>Answer (Ringng Call)</td>
<td>351</td>
<td>✓</td>
<td>✓</td>
</tr>
<tr>
<td>Automatic Intercom Access On/Off</td>
<td>361</td>
<td>✓</td>
<td>x</td>
</tr>
<tr>
<td>Automatic Line Access On/Off</td>
<td>360</td>
<td>✓</td>
<td>x</td>
</tr>
<tr>
<td>Automatic Line Answer</td>
<td>350</td>
<td>✓</td>
<td>x</td>
</tr>
<tr>
<td>Background Music On/Off</td>
<td>313</td>
<td>✓</td>
<td>x</td>
</tr>
<tr>
<td>Call Forward All Calls</td>
<td>355</td>
<td>✓</td>
<td>✓</td>
</tr>
<tr>
<td>Call Forward If Busy</td>
<td>357</td>
<td>✓</td>
<td>x</td>
</tr>
<tr>
<td>Call Forward If No Answer</td>
<td>356</td>
<td>✓</td>
<td>x</td>
</tr>
</tbody>
</table>
Table 12. ITP and SIP Features and Feature Codes (Continued)

<table>
<thead>
<tr>
<th>Feature Name</th>
<th>Feature Code</th>
<th>ITP Mode</th>
<th>SIP Mode</th>
</tr>
</thead>
<tbody>
<tr>
<td>Call Forward No Answer/Busy</td>
<td>358</td>
<td>✓</td>
<td>x</td>
</tr>
<tr>
<td>Conference</td>
<td>5</td>
<td>✓</td>
<td>✓</td>
</tr>
<tr>
<td>Default Station</td>
<td>394</td>
<td>✓</td>
<td>x</td>
</tr>
<tr>
<td>Directory</td>
<td>307</td>
<td>✓</td>
<td>x</td>
</tr>
<tr>
<td>Display Time And Date (Inter-Tel Protocol IP Mode)</td>
<td>300</td>
<td>✓</td>
<td>✓</td>
</tr>
<tr>
<td>Show IP Address (SIP Mode)</td>
<td></td>
<td></td>
<td></td>
</tr>
<tr>
<td>Do-Not-Disturb</td>
<td>370</td>
<td>✓</td>
<td>✓</td>
</tr>
<tr>
<td>Do-Not-Disturb Cancel</td>
<td>371</td>
<td>✓</td>
<td>✓</td>
</tr>
<tr>
<td>Do-Not-Disturb On/Off</td>
<td>372</td>
<td>✓</td>
<td>✓</td>
</tr>
<tr>
<td>Do-Not-Disturb Override</td>
<td>373</td>
<td>✓</td>
<td>x</td>
</tr>
<tr>
<td>Feature Button Default</td>
<td>395</td>
<td>✓</td>
<td>x</td>
</tr>
<tr>
<td>Group Listen</td>
<td>312</td>
<td>✓</td>
<td>✓</td>
</tr>
<tr>
<td>Handsfree On/Off</td>
<td>319</td>
<td>✓</td>
<td>x</td>
</tr>
<tr>
<td>Headset On</td>
<td>315</td>
<td>✓</td>
<td>✓</td>
</tr>
<tr>
<td>Headset Off</td>
<td>316</td>
<td>✓</td>
<td>✓</td>
</tr>
<tr>
<td>Headset On/Off</td>
<td>317</td>
<td>✓</td>
<td>✓</td>
</tr>
<tr>
<td>Hold – Individual</td>
<td>336</td>
<td>✓</td>
<td>✓</td>
</tr>
<tr>
<td>Hold – System</td>
<td>335</td>
<td>✓</td>
<td>x</td>
</tr>
<tr>
<td>Hunt Group Remove</td>
<td>322</td>
<td>✓</td>
<td>x</td>
</tr>
<tr>
<td>Hunt Group Replace</td>
<td>323</td>
<td>✓</td>
<td>x</td>
</tr>
<tr>
<td>Hunt Group Remove/Replace</td>
<td>324</td>
<td>✓</td>
<td>x</td>
</tr>
<tr>
<td>LCD Contrast Control</td>
<td>303</td>
<td>✓</td>
<td>✓</td>
</tr>
<tr>
<td>Message</td>
<td>365</td>
<td>✓</td>
<td>✓</td>
</tr>
<tr>
<td>Message – Cancel Message Left</td>
<td>366</td>
<td>✓</td>
<td>x</td>
</tr>
<tr>
<td>Message – Cancel Message On Endpoint</td>
<td>368</td>
<td>✓</td>
<td>x</td>
</tr>
<tr>
<td>Message – Silent Message</td>
<td>367</td>
<td>✓</td>
<td>x</td>
</tr>
<tr>
<td>Microphone Mute On/Off</td>
<td>314</td>
<td>✓</td>
<td>✓</td>
</tr>
<tr>
<td>Page</td>
<td>7</td>
<td>✓</td>
<td>x</td>
</tr>
<tr>
<td>Page Receive On/Off</td>
<td>325</td>
<td>✓</td>
<td>x</td>
</tr>
<tr>
<td>Program Buttons</td>
<td>397</td>
<td>✓</td>
<td>x</td>
</tr>
<tr>
<td>Program Station Password</td>
<td>392</td>
<td>✓</td>
<td>x</td>
</tr>
<tr>
<td>Queue (Callback) Request</td>
<td>6</td>
<td>✓</td>
<td>x</td>
</tr>
<tr>
<td>Record-A-Call</td>
<td>385</td>
<td>✓</td>
<td>x</td>
</tr>
<tr>
<td>Redial</td>
<td>380</td>
<td>✓</td>
<td>✓</td>
</tr>
</tbody>
</table>
## Table 12. ITP and SIP Features and Feature Codes (Continued)

<table>
<thead>
<tr>
<th>Feature Name</th>
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</thead>
<tbody>
<tr>
<td>Redirect Call</td>
<td>331</td>
<td>✓</td>
<td>✓</td>
</tr>
<tr>
<td>Reminder Message</td>
<td>305</td>
<td>✓</td>
<td>x</td>
</tr>
<tr>
<td>Reminder Message Cancel</td>
<td>306</td>
<td>✓</td>
<td>x</td>
</tr>
<tr>
<td>Remote Programming</td>
<td>359</td>
<td>✓</td>
<td>x</td>
</tr>
<tr>
<td>Reverse Transfer (Call Pick-Up)</td>
<td>4</td>
<td>✓</td>
<td>✓</td>
</tr>
<tr>
<td>Review Buttons</td>
<td>396</td>
<td>✓</td>
<td>x</td>
</tr>
<tr>
<td>Ring Intercom Always On/Off</td>
<td>377</td>
<td>✓</td>
<td>x</td>
</tr>
<tr>
<td>Ring Tone Selection</td>
<td>398</td>
<td>✓</td>
<td>✓</td>
</tr>
<tr>
<td>Routing Off</td>
<td>304</td>
<td>✓</td>
<td>x</td>
</tr>
<tr>
<td>Station Speed Dial</td>
<td>382</td>
<td>✓</td>
<td>x</td>
</tr>
<tr>
<td>Station Speed-Dial Programming</td>
<td>383</td>
<td>✓</td>
<td>x</td>
</tr>
<tr>
<td>System Forward Enable</td>
<td>352</td>
<td>✓</td>
<td>x</td>
</tr>
<tr>
<td>System Forward Disable</td>
<td>353</td>
<td>✓</td>
<td>x</td>
</tr>
<tr>
<td>System Forward On/Off</td>
<td>354</td>
<td>✓</td>
<td>x</td>
</tr>
<tr>
<td>System Speed Dial</td>
<td>381</td>
<td>✓</td>
<td>x</td>
</tr>
<tr>
<td>Switch Keymap</td>
<td>399</td>
<td>✓</td>
<td>x</td>
</tr>
<tr>
<td>Transfer To Hold</td>
<td>346</td>
<td>✓</td>
<td>x</td>
</tr>
<tr>
<td>Transfer To Ring</td>
<td>345</td>
<td>✓</td>
<td>x</td>
</tr>
</tbody>
</table>
ITP and SIP Feature Exceptions

Some endpoints cannot fully support certain features due to hardware or software restrictions. These exceptions are listed below by endpoint type.

Model 8600: The Model 8600 endpoint operates in ITP or SIP mode. However, it does not support the following features (feature code) in either mode:

- Feature Button Default (395)
- Microphone Mute On/Off (314)
- Program Buttons (397)
- Review Buttons (396)
- Switch Keymap (399)

Model 8690: The Model 8690 endpoint operates in ITP or SIP mode. However, the endpoint does not support the following features (feature code) in either mode:

- Change Language (301)
- Feature Button Default (395)
- LCD Contrast Control (303)
- Program Buttons (397)
- Review Buttons (396)
- Switch Keymap (399)

Models 8664, 8665, and 8668: The Model 8664, 8665, and 8668 wireless IP endpoints operate in ITP mode. However, they do not support the following ITP features (feature code):

- Automatic Intercom Access On/Off (361)
- Automatic Line Access On/Off (360)
- Background Music On/Off (313)
- Change Language (301)
- Group Listen (312)
- Handsfree On/Off (319)
- Headset On, Headset Off, Headset On/Off (315, 316, 317)
- LCD Contrast Control (303)
- Page Receive On/Off (325)
- Switch Keymap (399)

Model 8601: The Model 8601 operates in SIP mode only. However, it does not support the following SIP features (feature code):

- Headset On, Headset Off, Headset On/Off (315, 316, 317)
- LCD Contrast Control (303)
- Show IP Address (300)
**Feature Codes:** The two feature codes in Table 13 function differently, depending on the mode the endpoint is operating in.

**Table 13. Feature Codes with Different Meanings for ITP and SIP Modes**

<table>
<thead>
<tr>
<th>Feature Name</th>
<th>Feature Code</th>
<th>ITP Mode</th>
<th>SIP Mode</th>
</tr>
</thead>
<tbody>
<tr>
<td>Show IP Address Display Time/Date</td>
<td>300</td>
<td>Displays the system date and time, username, and extension.</td>
<td>Displays the IP address of the endpoint.</td>
</tr>
<tr>
<td>Show Version–Diagnostics–Show Version</td>
<td>9928 [9128 in Europe]</td>
<td>Displays the version and date of the call processing software.</td>
<td>Displays the firmware version on the endpoint.</td>
</tr>
</tbody>
</table>

**Voice mail displays:** In SIP mode, voice mail displays are not supported, however, the user will still receive audio prompts.

**IP Endpoint Flashing Light Indicators**

The endpoint Call buttons, trunk buttons, Speed Dial buttons, Hunt Group buttons, and some of the feature buttons contain light-emitting diodes (LED). LED flash rates indicate the status of the endpoints, trunks, and features.

For endpoints operating in ITP or SIP mode, Table 14 summarizes the meanings of the LED flash rates in interruptions per minute (IPM).

**Table 14. Flashing Light Indicators for IP and Multi-Protocol Endpoints**

<table>
<thead>
<tr>
<th>Button</th>
<th>Steady 0 IPM</th>
<th>Slow 30 IPM</th>
<th>Medium 120 IPM</th>
<th>Fast 240 IPM</th>
<th>Flutter 60/960 IPM</th>
</tr>
</thead>
<tbody>
<tr>
<td>ANSWER</td>
<td></td>
<td></td>
<td></td>
<td></td>
<td>Models 8620 and 8662: Intercom, recall, conference, camped on, or transferred call is ringing-in.</td>
</tr>
<tr>
<td>CALL</td>
<td></td>
<td>Call is active</td>
<td>Call is recalling from hold or transfer</td>
<td>Call or queue call-back is ringing in or camped on (waiting)</td>
<td>Call is on hold at the user’s endpoint or the user has entered a transfer feature code.¹</td>
</tr>
<tr>
<td>CNF</td>
<td>The LED flashes for 2–5 seconds and then goes out when the conference is established.</td>
<td>Conference is recalling your endpoint.</td>
<td></td>
<td>A conference party is on conference wait hold.</td>
<td></td>
</tr>
</tbody>
</table>
### Table 14. Flashing Light Indicators for IP and Multi-Protocol Endpoints (Continued)

<table>
<thead>
<tr>
<th>Button</th>
<th>Steady 0 IPM</th>
<th>Slow 30 IPM</th>
<th>Medium 120 IPM</th>
<th>Fast 240 IPM</th>
<th>Flutter 60/960 IPM</th>
</tr>
</thead>
<tbody>
<tr>
<td>DND</td>
<td>Endpoint is in Do-Not-Disturb.</td>
<td></td>
<td></td>
<td></td>
<td></td>
</tr>
<tr>
<td>FEATURE²</td>
<td>The feature is in use.</td>
<td></td>
<td></td>
<td></td>
<td></td>
</tr>
<tr>
<td>FWD</td>
<td>Calls are being forwarded.</td>
<td></td>
<td></td>
<td></td>
<td></td>
</tr>
<tr>
<td>HUNT GROUP</td>
<td>All stations in hunt group are unavailable.</td>
<td></td>
<td></td>
<td>Hunt group has a call camped on.</td>
<td></td>
</tr>
<tr>
<td>IC</td>
<td>Intercom call is in progress.</td>
<td>Recalling intercom call is camped on (waiting).</td>
<td>Intercom call is camped on (waiting).</td>
<td>An intercom call is on hold at the user’s endpoint or the user has entered a transfer feature code.¹</td>
<td></td>
</tr>
<tr>
<td>Mini-DSS²</td>
<td>The programmed endpoint is busy or unplugged.</td>
<td>The programmed endpoint is in DND.</td>
<td></td>
<td></td>
<td></td>
</tr>
<tr>
<td>MSG</td>
<td>Endpoint is in alphanumeric mode.</td>
<td></td>
<td>Message is waiting.</td>
<td></td>
<td></td>
</tr>
<tr>
<td>MUTE</td>
<td>Microphone is muted.</td>
<td></td>
<td></td>
<td></td>
<td></td>
</tr>
<tr>
<td>SPEAKER</td>
<td>Speakerphone is on or ready for use. Or, endpoint programming is in progress.</td>
<td></td>
<td></td>
<td>Model 8600: Intercom, recall, conference, camped on, or transferred call is ringing-in.</td>
<td></td>
</tr>
<tr>
<td>SPEED DIAL²</td>
<td>Designated endpoint is busy or unplugged.</td>
<td>Designated endpoint is in DND.</td>
<td></td>
<td>Designated endpoint has a call ringing in.</td>
<td>Continuous flutter indicates that the designated endpoint was left off hook.</td>
</tr>
</tbody>
</table>
For transfers, the IC or CALL button LED goes out when the user hangs up.

1. Requires a programmable button.

ITP and SIP Connectivity

Endpoints using ITP mode and SIP mode allow users to communicate using the Local Area Network (LAN) and the public telephone system.

IP and SIP calls can be routed through the Inter-Tel 5000 platform or they can be configured as members of the same Network Group. If the endpoints are configured as members of a Network Group, they can use the Peer-to-Peer (P2P) audio feature, which is explained on page 85.

**WARNING**

Possible Delay in Local Emergency Response to Remote Sites.

IP and SIP endpoint users should be alerted to the following hazardous situations:

- If an Emergency Call phone number is dialed from an IP or SIP endpoint located at a remote site that is not equipped with a correctly configured gateway, the call will be placed from the location where Inter-Tel 5000 Base Server is installed rather than from the location where the emergency call is made. In this situation, emergency responders may be dispatched to the wrong location. To minimize the risk of remote site users misdirecting emergency responders, Inter-Tel recommends regular testing of MGCP/SIP gateway trunk(s) for dial tone.

- If uninterruptible power supply (UPS) protection has not been installed as part of the Inter-Tel 5000 system, IP and SIP endpoints will not operate when electrical power fails either at remote sites or at the main system location.

To place calls during a power failure in this situation, IP and SIP endpoint users can only use a single line endpoint connected to one of the system’s power failure bypass circuits or a local endpoint not connected to the system.
The ports and connectors on IP and SIP endpoints vary depending on the model number. The following ports and connectors may be included on your endpoint:

- **Headset Jack**: Connects to a headset.
- **Handset Jack**: Connects to a handset.
- **Combined Headset/Handset Jack**: Connects to a headset or handset.
- **LAN/Power Jack**: Connects to an external power supply (see page 85).
- **LAN/Power Jack**: Connects to a network hub or a switch.
- **External Microphone Jack**: Allows you to connect an external microphone to the endpoint.
- **External Speaker Jack**: Allows you to connect an external speaker to the endpoint.
- **Reset Button**: Resets the connection to the network.
- **Compact Flash (CF) Port**: Allows you to connect a CF memory card if additional memory is needed.
- **Personal Computer Memory Card International Association (PCMCIA) Port**: Allows you to connect a PCMCIA memory card if additional memory is needed.
- **Universal Serial Bus (USB) Client Port**: Connects the endpoint to a host computer running ActiveSync®.
- **USB Host Port**: Allows you to connect a keyboard, mouse, or other USB device to your endpoint.
Battery Packs and Chargers

Models 8664 and 8665 require a charged battery pack to operate. The Nickel Metal Hydride (NiMH) rechargeable battery pack needs to be charged periodically. A charged battery pack provides four hours of talk time or 80 hours of idle time. To save battery power, instruct the user to place the endpoint in idle mode at the end of every call.

When the charge on the battery pack is low, the battery icon is displayed. The user will also hear a soft beep through the earpiece every six seconds when the battery is low. When the battery pack is critically low, the endpoint will not work until the battery pack is charged.

Power Supplies

Each SIP and IP endpoint must use a compatible power supply. Because there are many power supply options, contact your authorized Inter-Tel equipment provider when selecting a power supply for an IP or SIP endpoint.

IP Endpoint Limitations

SIP and IP endpoints:

- Do not have a secondary voice path and consequently cannot support the Off-Hook Voice Announce (OHVA) feature.
- Cannot use a PCDPM or MDPM. See “Digital Endpoint Data Port Modules” on page 102.
- Do not support full DSS/BLF units.
- Do not support Desktop Open Architecture Interface (OAI) applications.
- Do not support the Agent Help, Record-A-Call, and Station Monitor features while using P2P audio. See “System Features” on page 105.

Peer-to-Peer Audio for IP and SIP Endpoints

The P2P audio feature allows certain IP and SIP endpoints to transmit and receive audio directly with each other rather than through the Inter-Tel 5000 Base Server. The P2P feature reduces delay and packet loss for IP and SIP endpoints.

Endpoints that support P2P audio include:

- AudioCodes™ MP-104 MGCP gateway
- ITP endpoints, except the IP SoftPhone
- Multi-Protocol endpoints

Calls between IP and SIP endpoints that are members of separate Network Groups are routed through the Inter-Tel 5000 Base Server and do not use P2P.

The Network Group Diagnostics feature allows a system administrator to verify that the Network Groups are programmed correctly. For instructions on how to use the Network Group Diagnostics feature, see page 59.
Because IP endpoints connected on P2P calls do not use Inter-Tel 5000 resources, users cannot access Agent Help, Record-A-Call, and Station Monitor features.

**NOTICE**

Passing real-time streaming data, such as audio, through encrypted virtual private networks (VPN) may significantly impact the network performance, router and firewall functionality, and audio quality.

**Automated Boot Code Update**

**NOTICE**

**Severe endpoint damage possible.** If, during the boot code download, electrical current to the endpoint is interrupted (e.g., through disconnection or a power outage), the endpoint becomes inoperable and cannot be recovered. In that event, the endpoint must be returned to Inter-Tel for repair.

Information about the automatic update of endpoint boot code is included in this guide for the following reasons:

- You may be unaware that the update is occurring.
- Interruption of power to the endpoint during the download causes the endpoint to be inoperable and unrecoverable.

A Model 8620/8622 or 8662 endpoint with v2.0.0 or later firmware uses a TFTP server to update a crucial internal boot code. In any of the following situations, the endpoint checks its configuration file against a corresponding file on the TFTP server:

- On startup
- On command
- On periodic time-out

If the endpoint detects new parameters during this check, it automatically initiates the boot update procedure to download new boot code. However, the download is delayed if a call is in progress or music on hold is enabled. The endpoint display changes to alert you that the update is in progress.
Features Common to IP and Digital Inter-Tel Endpoints

Although their basic technologies and network connecting methods differ, IP and digital Inter-Tel endpoints share both some physical characteristics and software features. For example, both types of endpoint possess similar LCD windows, Call buttons, and feature buttons. They share a common data structure, data entry fields in DB Programming, and call processing features. The following sections describe important similarities of IP and digital Inter-Tel endpoints.

Direct Station Select/Busy Lamp Field Units

The Inter-Tel 5000 platform supports two 60-button DSS/BLF Units and one 50-button model. Although you can use either type of DSS/BLF Unit on any of the endpoints, the type of endpoint determines which is the more appropriate unit to use.

- **With Executive Display, Standard Display, Professional Display, and Associate Display Digital endpoints**: PC Data Port Module (PCDPM)-equipped Executive, Professional, Standard, and Associate Display endpoints may have up to three 60-button DSS/BLF Units connected to them. Up to twenty 60-key DSS/BLF Units can be installed on the system. Each unit is powered by a 450 mA AC transformer and connected to the PCDPM installed in the endpoint. The 6-rows-by-10-columns model, part no. 550.4200, is designed for Executive Display and Standard Display endpoints, and the 12-rows-by-5-columns model, part no. 560.4000, is designed for Professional and Associate Display endpoints.

- **With Model 8520 and Model 8560 endpoints**: PCDPM-equipped Model 8520 and Model 8560 endpoints support up to three Model 8450 50-button DSS/BLF Units, part no. 550.8450. Up to 20 of the 5-rows-by-10-columns DSS/BLF Units can be installed on the system. Each unit is powered by a 450 mA AC transformer and is connected to a PCDPM installed in the endpoint.

Model 8614 Mini-Direct Station Select Unit

The Mini-DSS Unit is programmable and provides one-button access to 16 extension numbers and/or feature buttons or resources. The Mini-DSS attaches to an IP Model 8660 or to digital models 8520 or 8560 and does not require a PCDPM or an AC transformer. Mini-DSS buttons contain LEDs that light or flash depending on the status of a station or the button’s programming. For details about the LEDs, see “IP Endpoint Flashing Light Indicators” on page 81.
Display Endpoint Screens

Several IP, SIP, and digital endpoint models are equipped with liquid crystal display (LCD) screens that show caller and feature information.

**NOTE**  For endpoints operating in SIP mode, voice mail displays are not supported, but the user does receive audio prompts.

**Six-Line Display**

Models 8660 and 8662 IP endpoints and the Model 8560 and Executive and Professional Display digital endpoints are equipped with a six-line LCD screens. See Figure 4 on page 89 and Figure 5 on page 89. Each line displays up to 16 characters. The top two lines display standard call information and show the same displays as other display endpoints. The remaining four lines display feature-related menus that change with the function being performed by the endpoint (intercom call, outside call, programming, etc.).

The lines have four menu selection buttons on each side of the screen (eight buttons) that are used to select displayed menu features. When two prompts appear on a line, such as PREVIOUS and NEXT, pressing the button next to the word selects the feature. When only one prompt appears on a line, such as DO NOT DISTURB, then pressing the button on either side of the prompt selects it.

A system-wide Alternate IP/Digital Endpoint Menu Displays flag determines the display that Model 8660 IP endpoint and Model 8560, Executive, and Professional digital endpoint users see during an outside call. In the default state, the display shows the following menu with Transfer-To-Hold on the first menu and Flash on the “More...” menu:

If the AlternateIP/Digital Endpoint Menu Displays flag is enabled, Model 8660 IP endpoint and Model 8560, Executive, and Professional digital endpoint users see the following menu with Flash on the first menu and Transfer-To-Hold on the More... menu:
Enabling the Alternate IP/Digital Endpoint Menu flag allows Centrex features to be accessed more easily, using the FLASH menu option.

**Figure 4.** *Model 8560, 8660, or 8662 LCD Screen*

Two-Line Display

The Model 8620/8622 IP endpoint, Models 8520 and Standard and Associate Display digital endpoints are equipped with a two-line displays that support 16 characters per line.

When the endpoint is not in use, the display shows the extension number, username, time of day, and date. Other displays include reminder messages, DND messages, numbers dialed, call sources, elapsed time of calls, current call costs, error messages, endpoint status, programming messages, etc. The displayed information is described throughout the instructions in this section of the guide.
**LCD Contrast Adjustment**

The contrast of the LCD screens on IP and digital display endpoints is adjustable. Eight different contrast levels are available. Examples follow for a two-line endpoint and a six-line endpoint. For instructions to adjust the contrast on a specific endpoint, refer to the User Guide, as identified in “Endpoint User Guides” on page 358.

Examples:

**To adjust the LCD contrast on the Model 8520, a two-line display endpoint:**

1. While the endpoint is idle, locate the contrast control wheel on the back, upper-right section of the endpoint.
2. Adjust the contrast of the LCD screen by turning the wheel clockwise (lighter) or counterclockwise (darker).

**To adjust the LCD contrast on the Model 8560, a six-line display endpoint:**

1. While the endpoint is idle, press the Special button ( ) and enter the LCD Contrast feature code (303).
2. Select LIGHTER or DARKER on the menu buttons.
   - OR, Use the Volume button to make your selection.
   - OR, Press the level (1–8) you want.
3. Press the Speaker button or ACCEPT to save your change. The Model 8620 IP endpoint is equipped with a contrast control knob to adjust the contrast level. This knob is located on the back of the endpoint and can be adjusted while viewing the display.

**Node and Circuit Number on Power-Up**

To aid in installation and troubleshooting procedures, display endpoints show their identification for 5 seconds when power is turned on, when the system is reset, or when the endpoint line cord is first plugged in (or removed and replaced). The extension number and assigned username appear on the top line of the display, and the node number and circuit number appear on the bottom line (for example, circuit 1.1.1 on node 2 would appear as 2:1.1.1). After the five-second identification display, the endpoint returns to the appropriate display, depending on its current status.
Default Inter-Tel Endpoint Feature Buttons

When the system is in the default state, all endpoints are in the same keymap group and have the following feature buttons. Feature code descriptions start on page 113. The Special button is signified with the infinity symbol (∞).

Table 15 summarizes the functions of endpoint feature buttons.

Table 15. Feature Button Functions

<table>
<thead>
<tr>
<th>Button</th>
<th>Function</th>
</tr>
</thead>
<tbody>
<tr>
<td>CALL BUTTONS 1–4</td>
<td>Selects an outgoing trunk, if idle, or answers a call on the button.</td>
</tr>
<tr>
<td>IC</td>
<td>Selects a voice channel or picks up a holding or camped-on intercom call.</td>
</tr>
<tr>
<td>DND</td>
<td>Programmable button 1: Do-Not-Disturb On/Off (372)</td>
</tr>
<tr>
<td>PROGRAMMABLE BUTTONS 2–6, 14–16, 18, and 21–35</td>
<td>Undefined values. Can be programmed with any feature code, trunk access code, or extension number, or can be programmed as an endpoint extension or System Speed Dial bin.</td>
</tr>
<tr>
<td>CNF</td>
<td>Programmable Button 7: Conference (5).</td>
</tr>
<tr>
<td>HOLD</td>
<td>Programmable Button 8: Individual Hold (336).</td>
</tr>
<tr>
<td>TRANSFER</td>
<td>Programmable Button 9: Transfer to Ring (345).</td>
</tr>
<tr>
<td>SYS SPD</td>
<td>Programmable Button 10: System Speed Dial (381).</td>
</tr>
<tr>
<td>REDIAL</td>
<td>Programmable Button 11: Redial (380).</td>
</tr>
<tr>
<td>PAGE</td>
<td>Programmable Button 12: Page (7).</td>
</tr>
<tr>
<td>QUEUE</td>
<td>Programmable Button 13: Queue Request (6).</td>
</tr>
<tr>
<td>MUSIC</td>
<td>Programmable Button 17: Background Music (313).</td>
</tr>
<tr>
<td>PAGE ZONE 1</td>
<td>Programmable Button 19: Selects Page Zone 1 for making a page.</td>
</tr>
<tr>
<td>STN SPEED DIAL</td>
<td>Programmable Button 20: Station Speed Dial (382).</td>
</tr>
<tr>
<td>FWD</td>
<td>Call Forward All Calls (355) Also used for canceling Call Forward.</td>
</tr>
<tr>
<td>SPECIAL</td>
<td>Signals the system to expect a feature code.</td>
</tr>
<tr>
<td>ANSWER</td>
<td>Answer (351).</td>
</tr>
<tr>
<td>OUTGOING</td>
<td>Selects a trunk for placing an outgoing call using the endpoint's Outgoing Call feature (8).</td>
</tr>
<tr>
<td>SPEAKER</td>
<td>Speakerphone On/Off.</td>
</tr>
<tr>
<td>MSG</td>
<td>Message (365) – Also retrieves messages when the flashing MSG button is pressed.</td>
</tr>
<tr>
<td>MUTE</td>
<td>Microphone Mute On/Off (314). This is a fixed-function button on digital endpoints</td>
</tr>
<tr>
<td>DATA</td>
<td>Data (340).</td>
</tr>
<tr>
<td>STATION SPEED DIAL BUTTONS 0–9</td>
<td>Used for storing and dialing one Station Speed Dial number each.</td>
</tr>
<tr>
<td>UNDEFINED</td>
<td>Some endpoints may have several undefined buttons that can be programmed as you want.</td>
</tr>
</tbody>
</table>
Keymaps

A keymap group determines the layout of the buttons for all the endpoints assigned to that group. If any of the buttons in the keymap group are user-programmable or undefined, they can be changed by the endpoint user. Up to 250 different keymap groups can exist. Each node contains its own keymap programming.

Each Inter-Tel endpoint is assigned a standard keymap and can also have an alternate keymap. For information about switching between standard and alternate keymaps, see page 156.

When the system is in the default state all endpoints are assigned to default Keymap 01. There are no individual preprogrammed keymaps for each of the specific endpoint types. And, the default Speaker, MSG, and MUTE buttons are not programmable.

As part of the installation procedure, the installer must make sure the buttons are labeled to match their functions.

The buttons in each keymap can be assigned the following functions:

- **Accept Button**: Not required for digital endpoints. When using numeric mode, digital endpoint users press on both arrows of the Volume button at once to accept an entry.

- **Cancel Button**: No further programming is necessary. This button is used to cancel a feature or cancel an entry when in numeric mode.

- **Call Button**: Up to 10 Call buttons can be assigned in each keymap. These buttons are used for call access. They are not associated with specific trunks. Instead, they are assigned to calls in the order that the calls are placed or received. In the following example, the user has three Call buttons.

  Table 16 describes the use of three Call buttons.

<table>
<thead>
<tr>
<th>Action</th>
<th>CALL 1 Button</th>
<th>CALL 2 button</th>
<th>CALL 3 button</th>
</tr>
</thead>
<tbody>
<tr>
<td>User places call #1</td>
<td>Shows status of call #1</td>
<td></td>
<td></td>
</tr>
<tr>
<td>Call #2 rings in while call #1 is active</td>
<td>Shows status of call #1</td>
<td>Shows ring flash for call #2</td>
<td></td>
</tr>
<tr>
<td>Calls #1 and #2 on hold, call #3 placed</td>
<td>Shows hold flash for call #1</td>
<td>Shows hold flash for call #2</td>
<td>Shows status of call #3</td>
</tr>
</tbody>
</table>

If there are more calls in progress than there are Call buttons (e.g., if another call rings in to the endpoint in the example), the new call will not have an available Call button and will camp on to the endpoint until a Call button becomes available. The call will then appear under the available button. Pressing an idle Call button will enter the programmed Outgoing Call feature for that endpoint. If no IC button is programmed in the keymap, intercom calls will appear under Call buttons.

**NOTE**: All keymaps should have at least one Call button if there is not an individual trunk button for every trunk.

- **Down Button**: No further programming is necessary. This button can be used scrolling backward through displays. When the endpoint is idle, it can be used for adjusting volume, in place of the Volume button. Digital endpoints use the Volume button.

- **DSS Button**: A DSS button dials the associated extension number when pressed. If assigned to a button with a lamp, it will also show the status of the associated station. Mini-DSS buttons cannot be changed by the station user.
• **Feature Button**: Programmed with feature codes. The feature code appears in the same button location on all endpoints assigned to the keymap and cannot be changed by the endpoint user.

• **Forward Button**: Can be programmed with any of the Call Forwarding feature codes. The forwarding code used by the button can be changed by the endpoint user.

• **Hunt Group Button**: Like DSS buttons, this button dials a hunt group pilot number when pressed and, if assigned to a button with a lamp, will show the status of that hunt group.

• **Individual Trunk Button**: This button is associated with one trunk. It can be used for placing outgoing calls, and for picking up calls on hold and answering incoming calls on that trunk.

• **Intercom (IC) Button**: The button allows access to intercom channels for placing and answering calls. If an button is not assigned, an intercom call will appear under a Call button.

• **Next Button**: Not required for digital endpoints. Digital endpoints use the Volume button for scrolling forward through displays.

• **Page Zone Button**: A page zone button is programmed with one of the page access codes (9600–9609 or 9600–9649 depending on the software version) for quick access to page zones (0–9 or 0–49 respectively).

• **Previous Button**: Not required for digital endpoints. Digital endpoints use the Volume button for scrolling backward through displays.

• **Programmable Button**: Up to 35 user-programmable buttons can be designated and given default values in the keymap database. They can then be reprogrammed to enter any of the trunk group access codes, extension numbers, or feature codes by the individual endpoint users as wanted. Or, they can be programmed as System or Station Speed Dial bins, or as secondary extension buttons (see below). The endpoint user or programmer can return the buttons to default values at any time.

• **Save Button**: Used for saving volume levels or programmed endpoint features. Digital endpoints use the Volume button, by pressing both ends at once.

• **Secondary Extension Button**: Allows an endpoint user to have one-button access to calls ringing or holding at another endpoint. It should be assigned to a button with a lamp so that the secondary extension button can show the call status. If necessary, an endpoint can have several secondary extension buttons, each associated with a different station. For details, see page 154.

• **SPKR Button**: Used for turning the endpoint speaker or headset on and off.

• **Speed Dial Buttons**: Each keymap can have up to 10 Speed Dial buttons. Each Speed Dial button can be programmed by the endpoint user to dial one digit string. If assigned to a button with a lamp, the lamp will show the status of an extension number if it is associated with that button.

• **Trunk Group Button**: These buttons are associated with specific trunk group access codes and are used for placing outgoing calls on trunks in the selected trunk group. If assigned to a button with a lamp, the lamp will be lit if all trunks in the trunk group are in use.
• **Undefined Button**: Any button can be left undefined if it will not be needed. It can be programmed, like a user-programmable button, as described above. If the endpoint user or programmer returns the endpoint to default values, the button will return to being “undefined.”

• **Up Button**: Not required for digital endpoints. When the endpoint is idle, the Up button can be used for adjusting volume, in place of the Volume button. Digital endpoints use the Volume button for scrolling forward through displays.

## Secondary Extension Appearances

The Secondary Extension Appearances feature allows calls ringing or on individual hold at one “primary” endpoint to appear on a “secondary extension button” at another endpoint. For example, a secretary or assistant can have a secondary extension button for a supervisor’s endpoint, or one endpoint can have several secondary extension buttons, each associated with a different primary endpoint. The primary endpoint and the secondary endpoint must reside on the same node.

A secondary extension button functions as follows:

• When a call is ringing or holding on any Call button at the primary endpoint, pressing the secondary extension button allows you to answer the call. This action allows you to answer recalls, transfers, callbacks, and hunt group calls.

• If the secondary extension button is unlit, pressing the button places an intercom call to the primary endpoint. If the button is flashing, an intercom call can still be placed without answering the incoming call by pressing the pound (`) button before the secondary extension button, by pressing a Mini-DSS Unit button, or by manually dialing the primary endpoint’s extension number.

Two programmable options may be used to determine the type of call indications the endpoint user receives for calls on a secondary extension button:

• **Ring When n Calls at Extension**: This programmable option allows the endpoint to receive a burst of ringing when “n” number of calls are present at the associated primary endpoint. The ring signal will repeat periodically as long as “n” number of calls are present at a primary endpoint. (This period is determined by the DKTS Secondary Extension Key Alerting Tone timer.) If “n” is set to 0, the secondary endpoint will never receive the burst of ringing. This option is programmed individually for each secondary extension button, during keymap programming.

• **Transient Call Indication on Call Answer**: This programmable endpoint flag determines whether the endpoint user will see a call display when answering a call by pressing the secondary extension button or reverse-transferring the call. If enabled, the display will tell if the call was ringing, recalling, transferred, or holding at the primary endpoint.

The secondary extension button can be located on an Inter-Tel endpoint, a Mini-DSS unit, or a DSS unit. There are two ways to make a secondary extension button:

• Create a secondary extension button in the keymap and determine which endpoint it will represent. This will be the same for all endpoints assigned to that keymap.

• Set the default value of a user-programmable button to “secondary extension button” in the keymap. This allows each endpoint user assigned to that keymap to determine the associated endpoint for his or her secondary extension button. For button programming procedures, see page 154.

User-programmable buttons cannot be made into secondary extension buttons by the endpoint user; they can only be created in the database. Also, the endpoint user cannot change the button into any other type of button (such as a Speed Dial or feature button), even though it is assigned to a user-programmable button.
Any Inter-Tel endpoint button can be designated as a secondary extension button, but buttons with lamps are recommended so that the secondary extension button can show call status. Table 17 on page 100 shows the definition for each possible lamp flash rate.

The secondary extension button shows only calls that appear under the Call buttons at the primary endpoint. Any calls on individual trunk buttons at the primary endpoint are considered private calls and will not appear on the secondary extension button.

Usually, the secondary extension button will not show intercom calls. However, if the primary endpoint does not have an IC button, intercom calls will appear under a Call button and, therefore, will appear on the secondary extension button. If the primary endpoint is a single line set, all calls will appear on the secondary extension button.

After answering a call on the secondary extension button, the endpoint user can transfer it back to the primary endpoint using one of the following methods:

- **Transfer to hold**: To transfer the call to hold at the primary endpoint, press the secondary extension button, announce the call if you want, and then hang up.
- **Transfer to ring**: To transfer a call to the primary endpoint, press the **TRANSFER** button and then the secondary extension button. Before hanging up to complete the transfer, you may announce the call.

Because the calls that appear on the secondary extension button are active only at the primary endpoint, they will not follow call forwarding or system forwarding paths at the endpoint with the secondary extension button.
Headsets

An electret headset can be connected to Inter-Tel Model 8600, 8620, and 8660 IP endpoints or to Inter-Tel Model 8500, 8520, and 8560 digital endpoints. Pressing the Speaker button connects or disconnects calls when the headset is attached.

Some headsets are equipped with a power-saver mode that disables the headset after a period of silence. To prevent these headsets from missing portions of incoming calls, the “Headset Connect Tone” endpoint flag must be enabled in DB Programming. Enabling this flag allows the endpoint to generate a tone that activates the headset before connecting to an incoming call. In the default state, the flag is disabled.

Models 8620 and 8660 IP endpoints and Models 8520 and 8560 digital endpoints are equipped with headset jacks on the back. Plugging into the jack allows you to have the headset and the handset connected at the same time. You must still, however, enable/disble the headset using the Headset Mode feature code.

Connecting a Headset

The following procedures describe how to connect a headset to an Inter-Tel endpoint.

To connect a headset to a Model 8620 or 8660 IP or a Model 8520 or 8560 digital endpoint:
1. Insert the headset plug into the vacant headset jack on the back of the endpoint.
2. While on-hook, enter the Headset On feature code (315) or the Headset On/Off feature code (317). The display shows HEADSET MODE ON.

To connect a headset to all other Inter-Tel endpoints:
1. Remove the coiled handset cord from the base of the endpoint. Leave the handset in the cradle.
2. Insert the headset plug into the vacant handset jack.
3. While on-hook, enter the Headset On feature code (315) or the Headset On/Off feature code (317). The display shows HEADSET MODE ON.
Disconnecting a Headset

The following procedures describe how to disconnect a headset from an Inter-Tel endpoint.

**To disconnect the headset from a Model 8620 or 8660 IP or a Model 8520 or 8560 digital endpoint:**
1. Unplug the headset cord from the back of the endpoint.
2. While on-hook, enter the Headset Off feature code (316) or the Headset On/Off feature code (317). The display shows HEADSET MODE OFF.

**To disconnect the headset from all other Inter-Tel endpoints:**
1. Unplug the headset cord from the base of the endpoint.
2. Plug the coiled handset cord into the base of the endpoint.
3. While on-hook, enter the Headset Off feature code (316) or the Headset On/Off feature code (317). The display shows HEADSET MODE OFF.

Digital Endpoints

With the addition of Digital Expansion Interface (DEI) equipment and appropriate licensing, the Inter-Tel 5000 Network Communications Solutions platform supports the following digital endpoints.

- Hard digital endpoints
  - Model 8500
  - Model 8520, two-line display
  - Model 8560, six-line display
  - Associate
  - Standard
  - Executive
  - Professional
  - Basic
- Single-Line Adapter (SLA)
- Single Line endpoints
- Modem Data Port Module (MDPM)
- Personal Computer Data Port Module (PCDPM), audio only
Inter-Tel Digital Endpoints

The following paragraphs describe the features supported by Inter-Tel endpoints. Throughout this guide, the term “Inter-Tel endpoint” refers to full-feature endpoints manufactured and sold by Inter-Tel. The term does not apply to industry-standard single line DTMF endpoints.

Display and Nondisplay Digital Endpoints

The following display and non-display digital endpoints have a built-in speaker and microphone that allow the user to answer intercom calls handsfree, dial while on-hook, monitor calls, and listen to background music.

- Models 8500, 8520, and 8560
- Executive Display
- Professional Display
- Standard Display
- Associate Display
- Axxess Basic Digital
- Eclipse Basic Digital

These endpoints can be programmed in individual endpoint programming to be allowed to use system resources for standard and enhanced speakerphones. Enhanced mode allows for full-duplex operation over a limited volume range (the range depends on trunk characteristics, typically the range is 1-3). Enhanced speakerphones are explained in the following section.

If a user presses the Speaker button while on-hook, the standard switched speakerphone function is activated and the speakerphone resource is assigned to that endpoint for the length of the call. If a user requests the speakerphone function when speakerphone resources are not available, the user will hear reorder tones, and the display will show NO SPKRPHN AVAIL STAY OFF HOOK.

If the user attempts to change to the speakerphone and hangs up during a call when speakerphones are unavailable, the user will be forced into a monitor-only mode. The user will be able to hear the other party but will not be allowed to transmit to them. In this mode, the user’s MUTE button will be lit, and the display shows NO SPKRPHN AVAIL COME OFF HOOK. Pressing the lit MUTE button will result in a burst of reorder tone; lift the handset to speak.

On certain handsfree-to-handsfree intercom calls, voice volume levels may cause feedback to occur. If this happens; one station user should pick up the handset. Two display endpoints cannot be in enhanced mode while on a call together; one endpoint must be off-hook to allow the other endpoint to enter enhanced speakerphone mode. If both endpoints are on-hook, there is no benefit to having one endpoint in enhanced mode.
Executive and Professional Display Speakerphones

Each Executive and Professional Display endpoint has a built-in, integrated speakerphone that allows on-hook dialing and handsfree operation on outside calls and inside (intercom) calls. The speakerphone can be used in standard mode or in enhanced mode which allows for full-duplex operation over a limited volume range (the range depends on trunk characteristics, typically the range is 1–3). Enhanced speakerphones are explained below.

The Executive and Professional Display endpoint’s speakers can also provide background music if an external music source is installed and the speaker is used to receive pages.

A programming option can be used to disable the endpoint’s speakerphone capabilities. Or, the user can disable the speakerphone for intercom calls using the Handsfree On/Off and/or Ring Intercom Always On/Off feature code.

On certain handsfree-to-handsfree intercom calls, voice volume levels may cause feedback to occur. If this happens, one station user should pick up the handset. Two display endpoints cannot be in enhanced mode while on a call together; one endpoint must be off-hook to allow the other endpoint to enter enhanced speakerphone mode. If both endpoints are on-hook, there is no benefit to having one endpoint in enhanced mode.

Enhanced Speakerphones for Digital Endpoints

The following display and nondisplay digital endpoints have enhanced speakerphone capabilities:

- Model 8500, 8520, and 8560
- Executive, Professional, Standard, and Associate Display
- Axxess and Eclipse Basic Digital

To use an enhanced speakerphone on a digital endpoint, the user establishes a call and presses $\text{Special}$ (the Special button) and then the Speaker button (or presses $\text{Special}$ and enters the Enhanced Speakerphone feature code, 310). There will be about one second of white noise at this point. Both parties should avoid speaking during this white noise, which is used to calibrate echo-canceling circuitry. After the white noise, there will be a confirmation tone that both parties will hear (except for Executive Display and Professional Display endpoint users who will hear the tone, but the other party will not). Then the user will be connected in an enhanced handsfree call. If the speakerphone cannot be calibrated correctly due to trunk conditions, the enhanced speakerphone will not be enabled and the display will show CANNOT ENABLE ENHANCED MODE.

It is normal for an outside party to hear a minimal amount of his or her own echo when speaking to a digital endpoint in enhanced speakerphone mode. This is usually acceptable to most callers on local calls and on most long-distance calls. However, on some T1 and long-distance calls involving significant delays (for example, cross-country T1 calls, overseas calls or calls via satellite), the amount of echo may be objectionable, and the enhanced speakerphone mode should not be used on these types of long-distance calls.

If an enhanced speakerphone call is placed on hold and then taken off hold on an Executive or Professional Display endpoint, it remains an enhanced speakerphone call. For all other digital endpoints, if the call is placed on hold, the call is returned to standard speakerphone mode.

A station can have only one enhanced speakerphone call. For example, if an Executive or Professional Display endpoint user makes an enhanced speakerphone call, puts it on hold, and establishes another enhanced speakerphone call, the first call will be turned back into a standard speakerphone call.
A digital endpoint user can set up a conference and then, once it is established, press the \( \text{Special button} \) and Speaker (or press \( \text{Special} \) and enter the Enhanced Speakerphone feature code, 310) to have an enhanced speakerphone call in the conference.

### Digital Endpoint Flashing Light Indicators

Table 17 summarizes digital endpoint LED flash rates in interruptions per minute (IPM).

<table>
<thead>
<tr>
<th>Table 17. Inter-Tel Digital Endpoint LED Flash Rates</th>
</tr>
</thead>
<tbody>
<tr>
<td><strong>Steady 0 IPM</strong></td>
</tr>
<tr>
<td><strong>CALL</strong></td>
</tr>
<tr>
<td><strong>CNF</strong></td>
</tr>
<tr>
<td><strong>DND</strong></td>
</tr>
<tr>
<td><strong>FWD</strong></td>
</tr>
<tr>
<td><strong>HUNT GROUP</strong></td>
</tr>
<tr>
<td><strong>IC</strong></td>
</tr>
<tr>
<td><strong>MSG</strong></td>
</tr>
<tr>
<td><strong>MUTE</strong></td>
</tr>
<tr>
<td><strong>SPEAKER</strong></td>
</tr>
</tbody>
</table>
Red and Green LEDs

Inter-Tel Models 8500, 8520, and 8560 digital endpoints; Executive, Professional, Standard, and Associate Display endpoints; and, Basic Digital endpoints have red and green LEDs in the button lamps. When both colors are used, a green lamp generally indicates that the call is active, ringing, or holding at a station (see below for a possible exception for direct ring-in calls). When the lamp is red, the call is active, ringing, or holding at another station. Trunk access, allowed-answer permission, and the feature being used (for example, System Hold or Call Transfer) determine whether you can pick up a (red) call from another station.

Two system-wide Database (DB) Programming options control these LEDs.

- **Allow Green LEDs**: This flag gives the installer the option of having all red/green endpoints and red-only endpoints operate the same, or allowing the red/green endpoints to use the green LEDs. If this flag is enabled, the dual-color endpoints will use both the red and green LEDs. If it is disabled, the endpoints will operate in red-only mode, like the endpoints without green LEDs. In the default state, the flag is disabled.

- **Use Green LEDs for Direct Ring**: This flag determines whether direct ring-in calls cause a red or green lamp. If the flag is enabled, direct ring-in calls to your station will be green. If disabled, all direct ring-in calls will cause the LED to be red. In the default state, this flag is enabled.

---

**IMPORTANT**
The Allow Green LEDs flag in DB Programming must be enabled. *Not applicable for programming multi-protocol IP endpoints.*

---

**Table 17. Inter-Tel Digital Endpoint LED Flash Rates (Continued)**

<table>
<thead>
<tr>
<th>SPEED DIAL or Mini-DSS</th>
<th>Steady 0 IPM</th>
<th>Slow 30 IPM</th>
<th>Medium 120 IPM</th>
<th>Fast 240 IPM</th>
<th>Flutter 60/960 IPM</th>
</tr>
</thead>
<tbody>
<tr>
<td>Trunk</td>
<td>Trunk is in use at another station, in a conference at your station, or is unplugged. (Red LED)</td>
<td>Trunk is in use at your station. (Green LED*)</td>
<td>Trunk is recalling from hold or transfer. (Red LED)</td>
<td>Trunk is ringing in or camped on (waiting). (Red LED if on another station or green if at your station*)</td>
<td>Trunk is on hold. (Red LED if on another station or green if at your station*)</td>
</tr>
</tbody>
</table>

* The LED changing color condition applies only to Inter-Tel digital endpoints equipped with LEDs that can light red or green in accordance with the way the buttons are programmed.
Digital Endpoint Data Port Modules

As described in the following paragraphs, two types of Data Port Modules that can be connected to Inter-Tel digital Models 8520 and 8560, Executive Display, Professional Display, Standard Display, and Associate Display endpoints.

Personal Computer Data Port Module

The optional Personal Computer Data Port Module (PCDPM) can be installed on Standard, Associate, Executive, and Professional Display endpoints or on Models 8520 or 8560 endpoints to gain Off-Hook Voice Announce (OVHA) receive capability or to install an optional Modem Data Port Module (MDPM).

When used in conjunction with the Inter-Tel 5000 DEI product, the PCDPM supports audio only with v1.1 and later software. In this version of software, the serial port is not functional on the PCDPM, and it does not transmit data, even though the name of the device suggests that it does.

Modem Data Port Module

The optional Modem Data Port Module (MDPM) can be installed on Standard Display, Executive Display, Model 8520, or Model 8560 endpoints for attaching a modem-equipped data device or a single line endpoint. The MDPM requires a PCDPM.

The Model 8416 Mini-DSS connects to Model 8520 or Model 8560 digital endpoints without PCDPM and MDPM devices.

Single Line Endpoints

Industry-standard, single line DTMF endpoints can be installed on the system and/or used as an off-premises extension (OPX). On-premises single line circuits can be used for installing AC-ringing single line endpoints, playback devices, and other single line devices.

Single line endpoints are connected to Single-Line Adapter (SLA) circuits, which are in turn connected to available DEM-16 circuits. Off-premises stations can be connected to SLAs or T1/E1/PRI modules.

- Each SLA can support any combination of up to two single line endpoint or DID devices.
- T1/E1/PRI module circuits can be used for installing off-premises single line endpoints.

Types of Single Line Endpoints

Industry-standard, single line endpoints can be installed on the system and/or used as an off-premises extension (OPX). An on-premises single line endpoint can be an AC-ringing single line endpoint, a playback device, and other single line device.

Single line dual-tone multi-frequency (DTMF) endpoint users gain access to system features by hookflashing (pressing and releasing the hookswitch quickly) and entering feature codes.
Off Premises Extensions (OPX)

Off premises extensions are remote single line DTMF endpoints connected to the Inter-Tel system through telephone company OPX circuits, or customer-provided circuits.

OPX endpoint users gain access to the features by hookflashing [recalling] (pressing and releasing the hookswitch quickly) and entering feature codes.

Some telephone company OPX networks do not recognize certain single line endpoint ring cadences used in the system. This incompatibility results in an inability to provide certain ring signals to OPX endpoints. Because of this, the system is equipped with an endpoint option known as Extended Ring Cadences, which allows the ring cadences be altered for individual single line circuits.

Extended Ring Cadences

Table 18 shows the different standard and extended ring cadences for the various types of single line endpoints used on the Inter-Tel telephone system in the United States.

Table 18. Ring Cadences for Single Line Endpoints

<table>
<thead>
<tr>
<th>Single Line Endpoint</th>
<th>Cadence Type</th>
<th>Description</th>
</tr>
</thead>
<tbody>
<tr>
<td>Single line endpoint connected to IP SLA, SLA, or T1 OPX</td>
<td>Standard IC Ring</td>
<td>A repeating 2-sec cadence that rings twice. Each ring lasts 200 ms and is separated from the next ring by 400 ms of silence. The remaining 1.2 sec is silent.</td>
</tr>
<tr>
<td></td>
<td>Extended IC Ring</td>
<td>A repeating 6-sec cadence that rings twice. Each ring lasts 800 ms and is separated from the next ring by 400 ms of silence. The remaining 4 sec are silent.</td>
</tr>
<tr>
<td></td>
<td>Standard CO Ring</td>
<td>A repeating 6-sec cadence that rings once for 2 sec followed by 4 sec of silence.</td>
</tr>
<tr>
<td></td>
<td>Extended CO Ring</td>
<td>Same as Standard CO Ring cadence.</td>
</tr>
<tr>
<td>Model 8560, Model 8520, Professional, Executive, Standard and Associate MDPM</td>
<td>Standard IC Ring</td>
<td>This cadence has two 250 ms rings every 2 sec.</td>
</tr>
<tr>
<td></td>
<td>Extended IC Ring</td>
<td>A repeating 3-sec cadence that rings once for 1 sec followed by 2 sec of silence.</td>
</tr>
<tr>
<td></td>
<td>Standard CO Ring</td>
<td>A repeating 6-sec cadence that rings once for 2 sec followed by 4 sec of silence.</td>
</tr>
<tr>
<td></td>
<td>Extended CO Ring</td>
<td>A repeating 6-sec cadence that rings once for 4 sec followed by 2 sec of silence.</td>
</tr>
</tbody>
</table>
Optional System Equipment

The following optional equipment can be purchased and installed with the Inter-Tel system:

**Voice Processing**: Depending on the selected voice processing option, provides multiport integrated voice processing features (e.g., Voice Mail, Automated Attendant, Hunt Group announcement and overflow, SMDR storage, Record-A-Call, etc.) Inter-Tel supports three types of voice processing systems. For more information about voice processing systems and supported features, see “External Voice Processing Systems” on page 22.

**Attendant Console**: Allows Attendants to use their personal computers (PC) to answer and manage incoming call traffic. It is primarily designed for use by Attendants of larger systems. However, due to its enhanced functionality, including electronic text messaging and directory dialing, the Attendant Console is an attractive option for smaller systems as well.

**External music sources**: External music sources (radios, tape players, etc.) can be connected to the system to provide music-on-hold and background music.

**External paging speaker equipment**: An amplifier and speakers can be connected to the system to provide external paging and background music to large areas, such as warehouses, loading docks, etc.

**Mini-DSS (Model 8416)**: The IP Model 8660 and digital Model 8560 support the Mini-Direct Station Select (Mini-DSS) Unit, which provides 16 programmable buttons that allow the user, typically an attendant, to monitor up to 16 extensions. The unit attaches directly to the Model 8660 endpoint and does not require any peripheral equipment.

**SMDR and error message recording output devices**: Output devices can be connected by IP socket only to record the data generated by the Station Message Detail Recording (SMDR) and error message recording features. The Inter-Tel 5000 platform does not support serial communication.

**Surge/spike protector**: A surge/spike protector should be installed to reduce the effects of AC voltage surges and spikes that may cause system malfunctions, false logic, and/or damage to the electronic components.

**Facsimile (fax) machine**: A fax machine may be connected and programmed to a single line port built into the Inter-Tel 5000 Base Server or connected with an IP-SLA or SLA port.

**Off-premises stations**: Single line DTMF endpoints can be connected to the system at a remote location. The connection is made through a telephone company OPX trunk.

**Playback devices**: A playback device such as an answering machine can be used in place of the voice processing system interface to help process calls. The device answers a call, plays a message, then disconnects from the call or transfers it to a Hunt Group to allow the caller access to other system resources.
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Introduction

The Inter-Tel 5000 Network Communications Solutions platform provides many user-friendly features. To describe the system features, this chapter has been divided into the following sections:

- **Access to the Features**: This section contains a list of the feature codes and their definitions.

- **System Organization**: This section describes possible administrator assignments, attendant arrangements, and hunt groups (including UCD and ACD hunt groups).

- **Trunk Features**: This section explains the trunk functions and programmable features, including: dual-tone multi-frequency (DTMF) or dial-pulse signaling; types of trunks (loop start, T1/E1/PRI); Caller ID, DNIS, and ANI; trunk groups; outgoing access, allowed-answer, and ring-in assignments; call routing tables; day and night modes of operation; automatic route selection (ARS); trunk and toll restriction; area/office code number plans; emergency calls; and direct inward system access (DISA).

- **Endpoints**: This section describes the endpoints supported by the system. Available endpoints include IP endpoints operating in Inter-Tel Protocol (ITP) mode, IP Single-Line Adapters (IP SLA), SIP endpoints, digital endpoints, and Mini-DSS (Direct Station Selection) Units, and single line endpoints such as single line endpoints, answering machines, and fax machines.

- **Intercom, Outside, and General Call Processing Features**: Several sections explain the use of the endpoint features that are available to all endpoint users, including intercom call features, outside call features, and general call processing features.

- **Attendant Features**: This section explains call processing features reserved for attendants and the use of Mini-DSS and DSS/BLF Units.

- **Administrator Features**: This section explains features reserved for administrators.

- **Record Keeping and Maintenance Features**: For record keeping purposes, the system offers station message detail recording (SMDR). A system error-reporting feature provides self-diagnostic information to make the system easier to service.

Types of Software Features

This section summarizes Inter-Tel 5000 software features—standard features that come with every system and premium features that are available at an additional cost and/or require additional licensing. Default feature codes for accessing system features appear in the Reference chapter—see “Default Feature Codes” on page 315.

Standard Features

The following standard features are included with each Inter-Tel 5000 platform:

- **Advanced Central Office (CO) Interfaces**: The advanced trunk features in this feature include Basic Rate Interface (BRI), Integrated Services Digital Network (ISDN), Primary Rate Interface (PRI) Private Network, and T1/E1/PRI.

- **Automatic Route Selection (ARS)**: ARS is a money-saving feature that allows the system to be programmed to select the least expensive route for placing outgoing calls. For details, see page 140.

- **Basic Call Processing Features**: The basic call processing features include standardized business features, such as Call Forwarding, Call Transfer, Call Waiting, Conference, Do-Not-Disturb (DND), Redial, Speed Dial, etc.
• **Basic Voice Mail (BVM):** Basic Voice Mail (BVM) provides basic voice messaging services for system users. BVM runs on the Linux® operating system and does not support Fax-On-Demand, buffered SMDR, or Unified Messaging. For additional details, see “Basic Voice Mail” on page 287 and “BVM Storage Capacities” on page 305.

• **Directories:** The Directory feature enables display endpoint users to “look up” intercom extension numbers and user names, System Speed Dial numbers and names, or system feature codes.

• **Multilingual Support:** Voice Processor prompts, endpoint displays, directories, DND messages, and reminder messages are available in American English, British English, Spanish, and Japanese. The database contains a programmable Language field for each trunk and endpoint to determine the default language. Also, a Change Language feature code can be entered by an endpoint user at any time to change the language.

• **Networking:** The Inter-Tel 5000 platform in any of its three configurations can operate as a stand-alone system or as a node in a network of up to 99 Inter-Tel Axxess® or Inter-Tel 5000 nodes. To support between 64 and 99 nodes, each node on the network must satisfy different requirements than those on a 63-node network. For details about supporting a 99-node network, contact your Inter-Tel equipment-and-service provider.

• **Record-A-Call:** This feature allows an endpoint user to record an ongoing call and place it in a voice mailbox.

• **System Forwarding:** System forwarding provides the ability to program the system so that calls ringing at an endpoint will follow a database-programmed “forwarding path” that routes the call based on the type of call and the status of the intended endpoint.

• **Uniform Call Distribution (UCD) Hunt Groups:** UCD hunt groups provide additional features to improve hunt group efficiency. When the UCD feature is enabled, additional system capabilities include announcement and overflow stations, hunt group priority ranking, and hunt group supervisors.

**Premium Features**

The following premium features are licensed by part number.

• **Additional T1/E1/PRI Ports:** Indicates how many additional Dual T1/E1/PRI port licenses are uploaded to the system.

• **Agent Help:** With Agent Help, an endpoint user can request help from a supervisor while on a call. The supervisor can then choose to monitor or join the call, or to reject the request.

• **Analog Voice Mail Hunt Groups:** A hunt group can be designated as an analog voice mail hunt group to enable it to contain the multiple ports of an analog voice mail unit (not Inter-Tel Voice Processing Units).

• **Basic Voice Mail Ports:** Indicates how many Basic Voice Mail port licenses are uploaded to the system.

• **Category A Endpoints:** Indicates how many Category A endpoint licenses are uploaded to the system. The following endpoints require a Category A license:
  – Models 8600, 8622, 8662, and 8690 operating in ITP mode
  – OASYS Net Phone, a third-party SIP endpoint

• **Category B Endpoints:** Indicates how many Category B endpoint licenses are uploaded to the system. The Model 8602 IP softphone requires a Category B license.
• **Category C Endpoints**: Indicates how many Category C endpoint licenses are uploaded to the system. The following endpoints require a Category C license:
  - Models 8600, 8622, 8662, and 8690 that are in SIP mode
  - Model 8660
  - Axxess/Eclipse IP PhonePlus
  - IP Single Line Adapter (SLA)
  - Models 8664, 8665, and 8668
  - Model 8601 SoftPhone for Pocket PC
  - Cisco® 7960 SIP Phone, a third-party SIP endpoint

• **Desktop Interface**: Desktop Interface is an Open Architecture Interface (OAI) feature that allows devices such as PC phones, operator consoles, ACD supervisors, and integrated contact management tools to be developed for the Inter-Tel 5000 platform.

• **Digital Expansion Interface #1**: Indicates whether or not the Digital Expansion Interface #1 software license is uploaded to the system.

• **Digital Expansion Interface #2**: Indicates whether or not the Digital Expansion Interface #2 software license is uploaded to the system.

• **Extended Node Capacity**: Indicates whether or not the IT-5000 License Enable 99 Nodes software license is uploaded to the system.

• **Inter-Tel Messaging IP Networking**: Indicates whether or not the Inter-Tel Messaging IP Networking software license is uploaded to the system.

• **IP Endpoints Enabled**: Indicates whether or not the IP Endpoints Enabled software license is uploaded to the system.

• **IP Gateway Trunks**: This feature determines the number of IP gateway trunks that can be connected to the system.

• **IP Networking**: Indicates how many IP networking trunk licenses are uploaded to the system.

• **Primary Rate Interface (PRI)**: PRI is an Integrated Services Digital Network (ISDN) version of a T1 circuit. A PRI circuit provides 23 bearer (B) channels and 1 data (D) channel (30B and 2D in Europe) and can transmit data at up to 1.544 Mbps (2.048 Mbps in Europe). To use PRI on the Inter-Tel 5000 platform, a T1/E1/PRI module is required in addition to the license. The license for this feature controls the number of PRI modules operate in the system.

• **Remote Automatic Call Distribution (ACD) Hunt Groups**: This feature enables members of ACD hunt groups to reside on different nodes. When using Agent IDs, each ID number must be treated as existing throughout the network, so the same Agent ID numbers must be established on all nodes on which hunt group members reside or log in. In order to use this premium feature, both this premium feature and the Automatic Call Distribution Hunt Groups premium feature are required for all nodes on which members reside or log in.

• **System Open Architecture Interface (OAI) Events**: System OAI Events is a system-level communications protocol that allows call management tools to be developed for the system, such as the Call Center Suite. Contact Inter-Tel for more information regarding these products.
• **System OAI Third Party Call Control (Level 2 OAI):** In addition to the call management tools allowed in System OAI Events, System OAI Third Party Call Control is a system-level communications TSAPI protocol that allows command-oriented tools to be developed for the system. Contact Inter-Tel for more information regarding these products.

• **Voice Processor AMIS Networking:** Voice Processing Units (VPU) can be connected in a network using analog Audio Messaging Interchange Specification (AMIS) protocol. Digital AMIS is not supported.

### Access to the Features

Inter-Tel endpoints have feature buttons that allow one-button dialing of feature codes. Information is programmed in the database to determine the arrangement of the feature buttons and their default values. As appropriate, some of the endpoint feature buttons can be designated as user-programmable buttons. See page 154 for more information.

#### Special Button and Hookflash [Recall]

Depending on the feature and system programming, users can either enter feature codes immediately after lifting the handset or while on-hook, or they must signal the system before entering the feature code. Inter-Tel endpoint users signal the system by pressing the Special button, which is represented by  on the endpoint. Single line DTMF endpoint users perform a hookflash [recall in Europe], by pressing then quickly releasing the hookswitch. If the user does not enter a code or begin dialing before the Dial Initiation timer expires, the system sends reorder/fast-busy tones.
Message Lamp

Inter-Tel IP endpoints have a Message Lamp located on the upper, right-hand corner. This lamp is programmable and is defaulted to the Message feature code, 365. If you change the feature code associated with this lamp, it flashes when the feature is active. For example, if you program the lamp for Individual Hold, feature code 336, the lamp flashes when a call is held at the endpoint.

Speaker Buttons

On all Inter-Tel endpoints, except the Model 8000 Series endpoints, handsfree-mode volume is controlled by a Speaker (SPKR) button. On Models 8600, 8620/8622, 8660, and 8662 endpoints, an illustration of a speaker (SPKR) represents the Speaker button.

Hold Buttons

On all Inter-Tel endpoints, except the Model 8000 Series endpoints, the Hold feature is controlled by a Hold (HOLD) button. On Models 8600, 8620/8622, 8660, and 8662 endpoints, the international hold illustration (Hold) represents the Hold button.

Dialpad “Hot” Buttons vs. “SPCL Key Required for Feature Code Entry” Flag

Unless programmed to use the Special button when entering a feature code, all Inter-Tel endpoints have dialpad “hot” buttons that allow dialing of feature codes, extension numbers, and trunk access codes without pressing the Special button or lifting the handset. As soon as a dialpad digit is pressed, the digit is recognized by the system.

There is a system-wide option called “SPCL Key Required for Feature Code Entry” that, if enabled, requires Inter-Tel endpoint users to always press the Special button before entering a feature code.

Regardless of programming, if an Inter-Tel endpoint user presses the Special button while on-hook, the display shows ENTER FEATURE CODE, and the Executive Display, Professional Display, or Model 8560 endpoint menu shows a list of feature codes. The user can enter the wanted code or press a menu button.

Feature Codes

Each of the endpoint features is assigned a feature code. Using the endpoint’s dialpad, these codes are entered to select trunks, process calls, and use special features. See the Reference section for details.

The codes are the values that are assigned when the system is in the default state. If wanted, they can be changed in Database (DB) Programming to any one- to five-digit value. Changing a feature code may affect the accessibility of other features.

- **Trunk Access Codes**: Used for selecting trunks when placing outgoing calls (see page 315).
- **Endpoint Feature Codes**: Used to access features available on endpoints (see page 316).
- **Administrator Feature Codes**: Used at Administrator endpoints to perform administrator functions, such as handling alarms and setting time or date (see page 322).
- **Diagnostics Feature Codes**: Used at Administrator endpoints to perform diagnostic functions, such as creating freezes and printing message logs (see page 324).
Extension Numbers

Extension numbers are recognized as feature codes by the system. When the system is in the default state, the extension numbers are assigned as shown in Table 19:

Table 19. Default Extension Numbers

<table>
<thead>
<tr>
<th>Function</th>
<th>Number</th>
</tr>
</thead>
<tbody>
<tr>
<td>Attendant</td>
<td>0</td>
</tr>
<tr>
<td>Endpoints</td>
<td>1000–1999</td>
</tr>
<tr>
<td>Hunt groups 1–75</td>
<td>2000–2074 or 2000–2299</td>
</tr>
<tr>
<td>Voice Processing System Applications</td>
<td>2500–2999</td>
</tr>
<tr>
<td>Page Zones</td>
<td>9600–9609 or 9600–9649 (Used only for programming a feature button to select a page zone.)</td>
</tr>
<tr>
<td>External Page Ports</td>
<td>91000–91003</td>
</tr>
<tr>
<td>Individual Trunks</td>
<td>94000–95999 (Determined by the bay and device number. Used for programming and troubleshooting.)</td>
</tr>
</tbody>
</table>

Attendant Endpoints

An attendant endpoint can be called by dialing 0 from any of the endpoints served by them. Typically, an attendant endpoint is programmed to provide the following services:

- Central operator for incoming calls
- Message center
- Recall endpoint for unanswered calls

Any endpoint can be designated as an attendant endpoint in DB Programming. Attendant stations should be equipped with display endpoints so the attendant can see the source names and numbers of Recall calls.

Digital endpoint Models 8520 and 8560 and IP endpoint Model 8660 support the Mini-DSS, which provides 16 user-programmable buttons. If equipped with Personal Computer Data Port Modules (PCDPMs), the following digital endpoints support 50-button or 60-button Direct Station Select (DSS)/Busy Lamp Field (BLF) units:

- Models 8520 and 8560
- Two-line and six-line display endpoints

Mini-DSS and DSS/BLF Units allow the attendant to visually monitor activity on stations, trunks, and other resources programmed under the lighted buttons.

A hunt group can be assigned to serve as an attendant. However, the individual stations in the hunt group are not required to be programmed as attendant endpoints, and the database will not reflect that the individual endpoints serve as an attendant for any other endpoints. The feature that separates functions is especially useful when a Voice Mail hunt group is assigned as an attendant. For information about Automated Attendant, see “Automated Attendant” on page 265.
Network and Local Primary Attendants

One attendant can be designated as the primary attendant who can receive unsupervised outside call recalls, hunt group recalls, and calls that cannot be matched to patterns in call routing tables.

When the Inter-Tel 5000 Network Communications Solutions platform is installed in a network, two types of primary attendants are possible:

- **Node Attendants**: There can be a primary attendant for each node.

- **Network Primary Attendant**: A network primary attendant can be programmed on each node. Typically, one network primary attendant is shared by all of the nodes. When the network needs to direct a call to an attendant, it attempts to direct the call to the network primary attendant first. If the network primary attendant is unavailable, it directs the call to the attendant on the node where the call originated.

If there is not a network or local primary attendant, calls that would normally go to the primary attendant (as described above) are handled as follows:

- If the system has seized the call, but it has not been sent to an endpoint, it is disconnected.

- If the call has been sent to an endpoint, it remains at the endpoint and rings until answered.

- If the call is not seized and not sent to an endpoint, the caller will hear ringing until he or she hangs up. The call will not ring at any endpoint.

One-Attendant or Multiple-Attendant Operation

Each node can be set up with one attendant or several attendants, as follows:

- **One Attendant**: One attendant provides all of the attendant services described in “Attendant Endpoints” on page 114. All trunks (except private trunks) are usually programmed to ring in at this attendant’s endpoint.

- **Multiple Attendants**: Any or all endpoints can be programmed as attendants. For example, one or more attendants may serve each department in a business. Trunks are programmed to ring at any or all attendant endpoints. Multiple attendants can be arranged in a hierarchy. That is, one attendant may be the attendant for another. In this case, the “serving” attendant is reached by dialing 0 at the “served” attendant.
Attendant Recall

When a call is placed on hold or is transferred from one endpoint to another, the Hold and Transfer timers limit the amount of time the call may remain unattended. After that time, the call recalls the endpoint that transferred it or placed it on hold, and the Recall timer is started. If the call remains unanswered at the endpoint until the Recall timer expires, it recalls the endpoint’s attendant and the Abandoned Call timer starts.

If there is no attendant, the call continues to recall at the endpoint that transferred it or placed it on hold. If the attendant endpoint is busy, the call camps on and the display shows the source of the recall. If the call is not answered before the Abandoned Call timer expires, the system disconnects the call.

If an attendant endpoint transfers a call and it is not answered before the Attendant Transfer timer expires, it will recall the attendant endpoint.

If an endpoint user transfers or forwards an outside call to an outside phone number, it becomes an “unsupervised” CO-to-CO call because no inside parties are involved. The CO-to-CO call is limited by the Unsupervised CO timer. When the timer expires, the call recalls the primary Attendant endpoint and causes the \textit{CNF} button to flash. Display endpoints show \textit{UNSUPERVISED CNF RECALL}. This serves two purposes:

- It allows the attendant to monitor the length of CO-to-CO calls. When a CO-to-CO call recalls, the attendant can disconnect the call or allow it to continue.
- If the callers hang up before the attendant receives the recall, the system may not have disconnected the trunks because a disconnect was not received from the central office. The attendant must disconnect the call.

When a recall rings at the attendant’s endpoint, a Call or individual trunk button flashes at the medium rate.

If the attendant has calls forwarded, recalls from endpoints follow internal call forward requests. Recalls do not forward to outside phone numbers, but recall the attendant’s endpoint until they are answered or the Abandoned Call timer expires. Placing the attendant’s endpoint in Do-Not-Disturb does not block recalls or direct ring-in calls.

If there is not a primary attendant, recalls remain at the endpoints and ring until answered or disconnected by the system.

To answer a hold or transfer recall at an attendant endpoint:

When you see a hold or transfer recall display \textit{<XXXX RCL FROM YYYY>} and hear a recall ringing:

1. \textit{If necessary}, lift the handset.

2. \textbf{Outside call recall}: Press the medium-flashing Call button or individual trunk button. Or, press the \textit{ANSWER} button. (If more than one trunk is recalling, pressing the \textit{ANSWER} button selects the outside call indicated on the display.)

   \textbf{Intercom call recall}: Press the \textit{IC} button or the \textit{ANSWER} menu button. (If you are busy when the intercom call recalls, it will camp on. The \textit{IC} button flashes at the medium rate, but you do not hear recall ring signals.)

   \textbf{Conference call recall}: Press the flashing \textit{CNF} button to connect with the conference call. The \textit{CNF} button flashes slowly and the display shows \textit{CONFERENCE IN PROGRESS}. \textit{If the parties are still talking}, press the \textit{CNF} button again and hang up to return the parties to their conversation. The \textit{CNF} button flutters. You can enter the conference at any time by pressing the fluttering \textit{CNF} button. If the Hold timer expires, the conference recalls your endpoint again. \textit{If the parties have hung up}, hang up to disconnect the call.
Hunt Groups

The Hunt Group feature permits calls to be placed to a group of stations and to be automatically transferred to an available endpoint within the group. Up to 75 hunt groups can be programmed in the database. Hunt group lists can contain individual endpoints and/or extension lists. Non-ACD hunt group stations must reside on the same node. Off-node devices must be ACD hunt group members.

The order in which hunt group endpoints receive incoming calls is determined by a list stored in the database. An endpoint or extension list can appear in a single hunt group more than once, and it can appear in multiple hunt group lists as needed.

Hunt groups have their own extension numbers (defaults to 2000–2074). Individual endpoints within the hunt group can be called using their assigned extension numbers.

Two other hunt group types are also available: UCD Hunt Groups, a standard feature on the system, and ACD Hunt Groups, which is a premium feature. For UCD information, see “UCD Hunt Groups” on page 121. For ACD, see “ACD Hunt Groups” on page 128.

- A DB Programming flag that enables pre-Record-A-Call messaging capability
- A flag titled “Return Automatic Call Distribution (ACD) Calls to Hunt Groups” that can be set for individual Hunt Groups (HG). When enabled, this feature requeues a call to the front of the Camp On queue for the HG the call came from. It allows a calling party to immediately return to the front of the HG queue if the assigned agent station goes into Do-Not-Disturb (DND) mode. If the flag is not enabled, the calling party will continue to ring until the No Answer Advance timer expires or the agent removes DND from the station and answers the call.
- A system-wide flag titled “Single Idle Time for All Hunt Groups.” For agent stations belonging to multiple hunt groups, this feature allows calls to be distributed to other stations having the longest idle time, regardless of a station’s idle time status in an individual hunt group.
Hunt Group Call Distribution

When an intercom or outside call is transferred or rings in to the pilot number, it circulates through the hunt group in linear or distributed order until answered, as described below.

- **Linear order**: Incoming calls always start circulating by ringing at the first endpoint (or extension list) on the hunt group list that is stored in the database.

- **Distributed order**: To even out the call load, distributed order shifts the starting point of each call. When an endpoint user receives a call, the next endpoint or extension list in the hunt group receives the next incoming call. If the receiving endpoint is busy (or all endpoints in the extension list are busy), or if there is no answer before the No Answer Advance timer expires, the call goes to the next endpoint or extension list on the hunt group list.

See “ACD Hunt Groups” on page 128 for information on the additional distribution types available in ACD hunt groups.

If an extension list is included in a hunt group, a call will ring all endpoints on the extension list when it reaches that point in the hunt group list. Therefore, to create an “all ring” type of hunt group, you can program the hunt group as either linear or distributed and then assign an extension list as the only hunt group member.
Hunt Group Call Processing

Hunt group endpoints receive the following indications when a call is ringing in:

- If an outside call is ringing, the endpoint designated to receive the call first shows ring flash on the associated individual trunk button (if it has one) or a Call button until the call is answered or the No Answer Advance timer expires and the call moves to the next endpoint.

- If hunt group camp on is allowed and all endpoints on the hunt group list are unavailable (busy, in Do-Not-Disturb, or with hunt group remove enabled), an intercom or outside call will camp on and the endpoints will receive the following indications:
  - **Busy endpoints** receive hunt group camp-on tones (if enabled) and display. If there is an individual trunk button associated with the trunk, it shows ring flash. (There is a programmable endpoint flag that can disable the camp-on tones for hunt group calls.)
  - **Endpoints with Hunt Group Remove enabled** receive camp-on tones and display. If there is an individual trunk button associated with the trunk, it shows ring flash.
  - **Endpoints with Do-Not-Disturb enabled** receive no camp on indications, but if there is an individual trunk button associated with the trunk, it shows ring flash.
  - **The first endpoint that becomes available** receives ringing and a flashing trunk or Call button. The camp-on and ring flash indications end at the other endpoints. (The associated trunk button will be steadily lit at those endpoints to show that the trunk is busy.)

- If hunt group camp on is not allowed, and all endpoints on the hunt group list are unavailable, the hunt group members will not have any camp-on indications. Callers will hear one of the following:
  - **Intercom callers** will hear busy signals.
  - **Callers on a trunk not needed to return busy signal** (such as a T1 E&M) will hear busy signals.
  - **Callers on a trunk needed to return busy signal** (such as a non-T1 loop-start) will hear ringing, and the call will be sent to the primary attendant.
  - **ISDN callers** hear busy signals.

Endpoints within the hunt group can receive direct trunk ring-in, intercom, forwarded, or transferred calls to their extension numbers without affecting other endpoints in the hunt group.
Hunt Groups and Call Forwarding

Hunt group programming affects the Call Forwarding feature in the following ways:

- **Hunt group calls follow unconditional forward**: If an endpoint in a hunt group is in the unconditional call forward mode to another endpoint, calls to the hunt group will follow the endpoint forwarding request. A forwarded hunt group call will ring at the forwarding destination until the No Answer Advance timer expires. If it is not answered before the timer expires, the call will return to the hunt group and continue circulating through the hunt group list.

- **Conditional forward dependent on timer interaction**: Endpoints that are busy and have their calls forwarded conditionally (no answer, busy, or unavailable) to another endpoint will receive the hunt group call (if the endpoint is not busy) until the Forward No Answer timer expires. At this point, if the hunt group’s No Answer Advance timer has not expired, then the call will be forwarded. After the No Answer Advance timer expires, the call will circulate to the next endpoint in the hunt group list.

- **Hunt group calls will not forward to some destinations**: Hunt group calls will not forward to Voice Mail, outside numbers, or system forwarding paths.

- **Announcement and overflow endpoints can forward hunt group calls**: If an announcement or overflow endpoint has call forward enabled, hunt group calls will follow the forward and the forwarding destination endpoint will act as the announcement or overflow endpoint.

- **Hunt groups can receive forwarded calls**: Endpoints can forward calls to a hunt group pilot number.

Hunt groups can be assigned as message centers and/or alternate message sources for individual endpoints.

**Hunt Group Remove/Replace and Do-Not-Disturb**

Hunt group members can temporarily stop hunt group calls from ringing at their endpoints by entering the Hunt Group Remove feature code as described below. (If an endpoint is assigned to more than one hunt group, this halts calls from all hunt groups. Hunt group assignments cannot be removed individually.) Do-Not-Disturb can also be used to halt hunt group and other calls to the endpoint (see page 238 for details).

This feature has no effect on ACD hunt groups. For ACD login/logout information, see page 129.

When the Hunt Group Remove feature is enabled, the user will still receive the Camp On display and tone, and the individual trunk button flashes, if one exists, for calls to the hunt group. The endpoint continues to receive calls placed to its extension number. Hunt group overflow and announcement stations cannot block hunt group calls using this feature.

If an endpoint user has programmed a button for entering the Hunt Group Remove/Replace feature code, and if that button has a lamp, the lamp will be lit whenever the endpoint is removed from the hunt group.
To remove or to replace an endpoint from its hunt group(s):

**Inter-Tel endpoints:** While on- or off-hook, enter the Hunt Group Remove feature code (322) to halt hunt group calls or enter the Hunt Group Replace feature code (323) to return the endpoint to its hunt group(s). Or, use the Hunt Group On/Off feature code (324) to toggle the hunt group mode on or off. You hear a confirmation tone, and the display shows ACCEPT (or DIVERT) HUNT GROUP CALLS. If off-hook, hang up.

**Single line endpoints:** Lift the handset and enter the Hunt Group Remove feature code (322) to halt hunt group calls or enter the Hunt Group Replace feature code (323) to return the endpoint to its hunt group(s). Or, use the Hunt Group On/Off feature code (324) to toggle the hunt group mode on or off. You hear dial tone.

**UCD Hunt Groups**

Uniform Call Distribution (UCD) hunt groups provide additional features to improve hunt group efficiency. The features added when UCD is enabled are as follows:

- Announcement and overflow endpoints pick up unanswered calls when the hunt group endpoints are busy.
- Hunt group priority ranking will place calls to one hunt group before another at endpoints that are members of more than one hunt group.
- Hunt group supervisors can monitor outside calls of any member of the hunt group.

**Announcement and Overflow Stations**

Announcement and overflow stations are individual endpoints that receive unanswered calls when all of the hunt group endpoints are unavailable. They are not usually included in the hunt group list. Announcement and overflow stations can be separate endpoints or they can be the same endpoint. Also, each of the hunt groups can have the same announcement and overflow stations or they can be assigned different endpoints. In a network, the announcement and overflow stations can be off-node devices.
Announcement Endpoints: When a call rings in or is transferred to a hunt group, the Announcement and Recall timers are started. The incoming call circulates through the hunt group until it is answered or the Announcement timer expires. When this timer expires, the unanswered call is picked up by the designated announcement endpoint which is a playback device that answers the call and plays a message. Meanwhile, the call continues circulating through the hunt group (unless it was sent to a voice processing system application and then transferred to an endpoint). If it is answered by an available hunt group endpoint while the announcement station is connected to the call, the call will leave the announcement station. Each direct ring-in call will transfer to the announcement endpoint only once. If it remains unanswered when the Overflow timer expires, the call will be sent to the overflow station. Or, if there is no overflow station, the call will be sent to the recall destination endpoint when the Recall timer expires.

Overflow Endpoints: The Overflow timer is started when the Announcement timer expires or, if there is no announcement endpoint, when the call is received by the hunt group. If an incoming hunt group call is unanswered when the Overflow timer expires, the call is picked up by an overflow station. The overflow station is a playback device that answers the call and plays a message.
Meanwhile, the call continues circulating through the hunt group (unless it was sent to a voice processing system application and then transferred to an endpoint). If the call is answered by an available hunt group endpoint while the overflow station is connected to the call, the call will leave the overflow station. The Overflow timer restarts each time the unanswered call leaves the endpoint at the overflow station. An unanswered call will return to the overflow station each time the Overflow timer expires, until the call is answered by a hunt group station or it is sent to the recall destination station.

The following table shows the path that an incoming hunt group call follows for possible combinations of announcement and overflow stations:

**Table 20. Incoming Hunt Group Call Paths**

<table>
<thead>
<tr>
<th>Announcement Station</th>
<th>Overflow Station</th>
<th>Call Path</th>
</tr>
</thead>
<tbody>
<tr>
<td>No</td>
<td>No</td>
<td>Call remains in hunt group.</td>
</tr>
<tr>
<td>Yes</td>
<td>No</td>
<td>Call goes to announcement endpoint only once, after Announcement timer expires.</td>
</tr>
<tr>
<td>No</td>
<td>Yes</td>
<td>Call goes to overflow endpoint after each expiration of Overflow timer.</td>
</tr>
<tr>
<td>Yes</td>
<td>Yes</td>
<td>Call goes to announcement endpoint once (after Announcement timer expires) then goes to overflow endpoint after each expiration of Overflow timer.</td>
</tr>
</tbody>
</table>

If an announcement or overflow station has Call Forward enabled, hunt group calls will follow the forward, and the forwarding destination endpoint will act as the announcement or overflow station.

**Voice Processing System Applications as Announcement/Overflow Stations**

When a voice processing system application is used as an announcement or overflow station, calls to the application automatically stop circulating through the hunt group if the caller selects a valid digit translation option (such as pressing a digit to select an extension from a Call Routing Announcement menu), and the call is transferred. This allows the application to send the call to other endpoints without the call being “pulled back” into the hunt group when a hunt group endpoint answers the call. However, if the caller does not dial a valid digit translation option, the call will be pulled back if a hunt group member answers.

Standard Display, Associate Display, Axxess Basic, Eclipse Basic, Model 8500, 8520, and 8560 endpoints do not send DTMF tones on intercom calls and therefore cannot stop circulating through the hunt group by dialing a digit.

A Call Routing Announcement application’s message can be programmed to include the caller’s queue position and/or estimated wait time. The queue position announcement tells the caller how many calls are ahead of his or her call. This includes calls being served and waiting calls (however, all calls being served count as one call). The estimated wait time is based on a programmed Average Connect Time Per Call multiplied by the number of calls ahead of the caller in the queue, divided by the number of available hunt group members \((\text{avg. connect time per call} \times \text{no. of waiting calls} \div \text{available members})\). For details, see “Call Routing Announcement” on page 272.
If a voice processing system application is used as an announcement or overflow endpoint or as the recall destination, and the system is unable to communicate with the voice processing system, outside calls will not be sent to the announcement or overflow application. They will continue to camp on to the hunt group.

**UCD Hunt Group Priority List**

Some endpoints may be members of more than one UCD hunt group. For this reason, hunt groups are assigned a “priority level.” The priority level determines which hunt group’s calls should be received first when calls ring in or camp on to several hunt groups at once.

UCD priority levels range from 0 (low) to 75 (high). If an endpoint is a member of multiple hunt groups that have the same priority level, calls received by those hunt groups will be queued in the order they were received by the telephone system. (Non-UCD hunt groups are treated as if they have a priority level of 0.)

**Figure 7. UCD Hunt Group Priority Levels**

<table>
<thead>
<tr>
<th>PRIORITY LEVEL</th>
<th>ORDER CALLS ARE RECEIVED</th>
</tr>
</thead>
<tbody>
<tr>
<td>0</td>
<td>Ext. 2000</td>
</tr>
<tr>
<td>1</td>
<td>Ext. 2001</td>
</tr>
<tr>
<td>2</td>
<td>Ext. 2002</td>
</tr>
</tbody>
</table>

**UCD Hunt Group Supervisors and Station Monitoring**

The Station Monitor feature allows hunt group supervisors to monitor the calls of anyone in a specified hunt group. It can be useful in training or in evaluating the performance of hunt group members.

As a courtesy, hunt group members should be notified in advance that their calls may be monitored. In addition, a programmable option can be enabled that sends a tone to the endpoint being monitored whenever the hunt group supervisor joins an ongoing call.

Call monitoring may be illegal in some locations. The end user is responsible for ensuring that use of this feature is in compliance with local law.

In DB Programming, each hunt group can have one or more endpoints assigned as the hunt group supervisor(s). The supervisor must reside on the same node as the hunt group members. (An extension list can be assigned as the supervisor to provide multiple supervisors.) The supervisor is usually not a member of the hunt group. If the supervisor is a member of the hunt group, the Hunt Group Remove/Replace feature can be used at any time without affecting the Station Monitor ability. If appropriate, one endpoint can be assigned as the supervisor for more than one hunt group.
To monitor a hunt group member’s call, the supervisor enters the Station Monitor feature code (321) and dials an extension number. The supervisor is then connected to the call and can hear both parties, but cannot be heard by either one. If the monitored call is terminated, transferred, or placed on hold by the hunt group member, the monitor function is terminated.

In the associated hunt group, the supervisor may monitor any active intercom or CO-to-intercom call (both hunting and non-hunting), including incoming, outgoing, and DISA-to-intercom calls. Conference calls and calls that do not involve hunt group members cannot be monitored.

If the supervisor attempts to monitor an endpoint that is not on an active call that allows monitoring, the system sends reorder tones, and the supervisor must enter the feature code again to try another number. If the supervisor attempts to monitor a non-hunt group or an idle endpoint in the hunt group, the system sends reorder tones and cancels the Endpoint Call Monitor feature.

Station monitoring requires conferencing circuits. If resources are not available when a supervisor attempts to monitor an endpoint, the supervisor’s display will show NO CNF CIRCUITS AVAILABLE, and the monitor will not be allowed. Multiple supervisors can monitor the same endpoint, providing that a conference circuit is available for each supervisor.

The supervisor endpoint cannot use the Agent Help feature while monitoring a call.

Station Monitoring is not supported on an IP or SIP device engaged in a P2P call. For details, see “Peer-to-Peer Audio for IP and SIP Endpoints” on page 85.

To monitor a hunt group call (designated supervisor only):

1. **To use the speakerphone:** While on-hook, enter the Endpoint Call Monitoring feature code (321). You hear a confirmation tone and the Speaker button lights. The display shows ENTER EXT TO MONITOR.

2. **To use the handset:** Lift the handset and enter the Endpoint Call Monitoring feature code (321). You hear a confirmation tone. The display shows ENTER EXT TO MONITOR.

3. **To monitor another hunt group member’s call:** Enter the Endpoint Call Monitoring feature code (321), and dial the extension number or press the Speed Dial button.

   **To terminate monitor:** If off-hook, hang up. If on-hook, press the Speaker button.

### Supervisor Barge-In

The following sections describe Barge-In enhancement to the Station Monitor feature and how to access Barge-In functions.

### Station Monitor Enhancements

The Barge-In feature adds new functionality to the Station Monitor feature. When silent monitoring a call, the supervisor now has the following options:

- **Barge-In:** Allows the supervisor to barge-in on a call to help the hunt group member/agent (default feature code is 386).
- **Record:** Allows the supervisor to record the call to review it later (feature code 385). The supervisor may also hang up and continue to record the call until one of the parties on the call hangs up or puts the call on hold. The record option allows the supervisor to record several calls at once and have them delivered to his or her mailbox.
• **Steal**: Allows the supervisor to steal (take away) the call from the hunt group member/agent (default feature code is 387).

• **Join and Record**: (For 6-line display endpoints only.) Allows the supervisor to join and record the call simultaneously. This feature is useful if the supervisor wants to review the call later.

Supervisors must be monitoring a call before they can access any of the Barge-In features (barge-in, steal, record, or join and record). Like the Station Monitor feature, Barge-In requires conferencing circuits. If resources are not available when a supervisor attempts one of these features, the system sends reorder tones and the supervisor sees NO CNF CIRCUITS AVAILABLE on the display.

A silent monitor call is rejected if:

• A hunt group member or agent is on a conference call.
• The hunt group member or agent is on a peer-to-peer (P2P) call.
• A hunt group member or agent is getting agent help.
• Another supervisor has already joined the call.

When the above situations occur, the supervisor will see SILENT MONITOR REJECTED on the display.

The Barge-In/Silent-Monitor feature can be used across all Inter-Tel 5000 and Axxess network nodes, as long as all Axxess nodes are running at least v9.0. The system administrator can also add off-node Endpoints as hunt-group supervisors.

**Accessing the Barge-In Features**

A supervisor may silent monitor a hunt-group member/agent if that supervisor is designated as a supervisor of the group to which the member/agent is logged in.

Three fields are associated with the Barge-In feature in DB Studio. These programmable fields control the Barge-In notification tone and display.

**To monitor a call:**

1. **To use the speakerphone**: While on-hook, enter the Station Monitor feature code (default is 321). After hearing a confirmation tone, the Speaker button lights. The display shows ENTER EXT TO MONITOR.

2. **To use the handset**: Lift the handset and enter the Station Monitoring feature code (default is 321). After a confirmation tone, ENTER EXT TO MONITOR is displayed.

3. **While silent monitoring a call**, you have the following options:
   - Barge in on the call
   - Return to silently monitoring the call
   - Record the call.

**NOTE** Only 6-line display endpoints have the menu buttons mentioned in these instructions. Users who have endpoints with a 2-line display, or no display, must use the feature codes to access the Barge-In features.
**To barge in on a call:** Press the **BARGE-IN** menu button or enter the feature code 386 (default). On display endpoints, the display shows **BARGE-IN IN PROGRESS**. While on a call, you can monitor, record, or steal the call by entering the appropriate feature code or by pressing the appropriate menu button.

**To return to silently monitoring the call:** Press the feature code 386 (default) again or press the **MONITOR** menu button.

**To record the call:** Press the **RECORD** menu button or enter the Record-A-Call feature code (default is 385). On display endpoints, the display shows **MONITORING <extension number>**. While on a call, enter the appropriate feature code or press the appropriate menu button to monitor or steal the call.

**To stop recording the call:** Enter feature code 385 again or press the **STOP RECORDING** menu button.

**To steal a call:** Press the **STEAL** menu button or enter the Steal feature code (default is 387). On display endpoints, the display shows **CALL STOLEN FROM <extension number>**. The hunt group member display shows **CALL STOLEN BY <extension number>**. When the supervisor steals a call, it becomes a regular call. While on the call, enter feature code 385 or press the **RECORD** menu button to record it.

**To join and record a call:** Press the **JOIN AND RECORD** menu button or enter the Join feature code (default is 386), followed by the Record-A-Call feature code (default is 385). On display endpoints, the display shows **BARGE-IN IN PROGRESS**. While on a call, the supervisor can monitor, stop recording, or steal the call by entering the appropriate feature code or press the appropriate menu button.

**To stop recording the call:** enter feature code 385 (default) again or press the **MONITOR** or **STOP RECORDING** menu button.

4. **To terminate any of the features:** If **off-hook**, hang up or enter the specific feature code of the feature you want to terminate. If **on-hook**, press the Speaker button or enter the specific feature code of the feature you want to terminate. On display endpoints, the display shows **BARGE-IN TERMINATED** or **SILENT MONITOR TERMINATED** (if you are silent monitoring at the time of termination).
ACD Hunt Groups

**IMPORTANT** The Automatic Call Distribution Hunt Groups premium feature is required to use ACD hunt group features.

Automatic Call Distribution (ACD) can be programmed to distribute hunt group calls to equalize call time or call count among the available members. If the System OAI Events premium feature is included in your software license, ACD hunt groups can also be programmed to send call information records that can be processed by an external device connected to a system serial port (such as Contact Center Suite).

ACD hunt groups can use the standard hunt group features described on page 117, and/or the UCD features described on page 121.

**ACD Call Distribution**

When an intercom or outside call is transferred or rings in to the ACD hunt group, it can circulate in linear or distributed order (as described on page 117) or using one of the following ACD distribution methods:

The Restart ACD Idle Time Upon Login flag affects how the longest idle time is calculated. See page 131 for more information.

- **Longest Idle**: An incoming call is sent to the endpoint that has not been involved in a call to this hunt group for the longest period of time. (It does not count calls that were received through other hunt groups, direct ring-ins, or transfers.)

- **Balanced Call Count**: To balance the call load, each incoming call is sent to the endpoint that has received the fewest calls through this hunt group. (It does not count calls that were received through other hunt groups, direct ring-ins, or transfers.)

When an agent logs in to a balanced call count ACD hunt group (or comes out of Do-Not-Disturb mode), the system checks the average call count for that ACD hunt group and assigns that average count to the endpoint logging in. This prevents the agent who is logging in from receiving all of the incoming calls until the call count of the lowest agent catches up with his call count. For example, if there are five agents logged in whose call counts are 5, 5, 5, 10, and 10, the average call count of 7 will be assigned to the next agent who logs in.

If an extension list is included in an ACD hunt group set for Longest Idle or Balanced Call Count distribution, it will treat each endpoint in the extension list as a separate agent; it will not ring all of the endpoints on the list at the same time. If the hunt group is set for linear or distributed order, a call will ring all endpoints on an extension list at the same time when the call reaches that point in the hunt group list.
ACD Agent IDs

ACD hunt group members are referred to as “agents.” Agents log in to the ACD hunt group to receive calls and log out to halt ACD hunt group calls. An ACD hunt group can be programmed to circulate calls to agents in two ways:

- **Agent IDs**: If the hunt group is programmed to use ACD Agent IDs, each agent is assigned an Agent ID number which he or she enters during the login procedure (described below). The hunt group calls are routed to logged in agents, according to their Agent ID number instead of their endpoint extension. Because the Agent ID is not associated with any endpoint extension, the agent can use any endpoint in the system to log in and does not have to use the same endpoint every time.

- **Members**: If the hunt group is not programmed to use Agent IDs, it will have a list of endpoints and will send calls to the endpoints where agents are logged in.

**IMPORTANT** For optimum system performance, there should not be more than 1000 Agent IDs in any hunt group and no more than 2000 Agent ID entries in all hunt groups combined.

ACD Agent IDs can be included in Extension Lists which allow several ACD Agents to receive a call at once.

When a call camps on to an ACD hunt group that uses Agent IDs, only the agents currently logged in to the hunt group will receive camp-on indications. ACD hunt group supervisors will receive visual Camp On displays if they are programmed as members of the hunt group and have the ACD Agent Logout feature enabled.

ACD Agent Log In and Log Out

ACD hunt group members are referred to as “agents.” Agents can log in to and out of the ACD hunt group at any time. While logged in, the agent will receive calls through the ACD hunt group. When the agent is logged out, calls to that ACD hunt group will bypass the endpoint. (The Hunt Group Remove/Replace feature, described on page 120, does not have any effect on ACD hunt group calls when entered by an ACD agent. However, placing the agent’s endpoint in Do-Not-Disturb will halt all ACD and non-ACD hunt group calls.)

Figure 8 illustrates how the Agent ID is used to log in or log out.

**Figure 8. ACD Hunt Group Agent Log In and Log Out**
There are three feature codes that can be used for logging in to and out of the ACD hunt groups: ACD Agent Login, ACD Agent Logout, and ACD Agent Login/Logout. The first two perform only one operation. The third (Login/Logout) is a toggle feature code that logs the endpoint in or out of all associated ACD hunt groups at once. If the toggle feature code is programmed in a user-programmable button with a lamp, the lamp will be lit when the agent is logged in to an ACD hunt group and available to receive calls, and it will be unlit when the agent is logged out.

To log out of all of your ACD hunt groups at once using the login/logout feature code:

While on- or off-hook (single line users must lift the handset), enter the ACD Agent Login/Logout feature code (328). The display shows AGENT LOGGED OUT OF ALL ACDS, and you hear a confirmation tone. Hang up if off-hook.

To log in to all of your ACD hunt groups at once using the login/logout feature code:

1. While on- or off-hook (single line users must lift the handset), enter the ACD Agent Login/Logout feature code (328). If you were already logged in, the display shows AGENT LOGGED OUT OF ALL ACDS, and you hear a confirmation tone. If you were not logged in, continue to the next step.

2. The display shows AGENT LOGIN AGENT ID. Do one of the following:

   - To log in to all ACD hunt groups for which you have an Agent ID: Enter your Agent ID. The display shows AGENT LOGGED INTO ALL ACDS. You are logged into only the ACD hunt group(s) that use the Agent ID that you entered. Hang up if off-hook. If another agent is already logged in at this endpoint, you hear reorder tones, and the display shows DIFFERENT AGENT ID ALREADY USED. You must have the other agent log out before you can use that endpoint.

   - To log in to all of your ACD hunt groups that do not use Agent IDs: Press #. The display shows AGENT LOGGED INTO ALL ACDS. You are logged into only the ACD hunt group(s) that do not use Agent IDs. Hang up if off-hook.

To log in to one or more hunt groups using the ACD agent login feature code:

1. While on- or off-hook (single line users must lift the handset), enter the ACD Agent Login feature code (326). The display shows AGENT LOGIN ACD NUMBER.

2. Enter the pilot number of the ACD hunt group you want. (Or, you can press # to log in to several ACD hunt groups at once.)

   If you entered an invalid hunt group number, the display shows NOT AN ACD HUNT GROUP, and you hear reorder tones.

3. The display shows AGENT LOGIN AGENT ID. Then do one of the following:

   - To log in to ACD hunt group(s) using Agent IDs: Enter your Agent ID. The display shows AGENT LOGGED INTO ALL ACDS. You are logged into only the ACD hunt group(s) that you requested in step 2 that use the Agent ID that you entered in this step. If another Agent ID is already logged in at this endpoint, you hear reorder tones and the display shows DIFFERENT AGENT ID ALREADY USED. You must have the other agent log out before you can use that endpoint.

   - To log in to ACD hunt group(s) that do not use Agent IDs: Press 7. The display shows AGENT LOGGED INTO ALL ACDS. You are logged into only the ACD hunt group(s) that you requested in step 2 that do not use Agent IDs.

   If you are already logged in, the display shows ALREADY LOGGED INTO XXXXX.

   If you are not a member of the entered hunt group, the display shows NOT AN AGENT IN THIS HUNT GROUP, and you hear reorder tones. Start over.

4. If off-hook, hang up. Repeat the procedure to log in to additional ACD groups.
To log out of an ACD hunt group:

1. While on- or off-hook, enter the ACD Agent Logout feature code (327). One of the following will occur:

   *If you were logged in to only one hunt group,* you hear a confirmation tone, and the display shows AGENT LOGGED OUT OF <hunt group>. The procedure is complete. Hang up if you were off-hook.

   *If you were logged in to more than one hunt group,* you hear a confirmation tone, and the display shows AGENT LOGOUT ACD NUMBER. Go to the next step.

   *If you were not logged in to any hunt group,* the display shows NOT LOGGED INTO ANY ACD GROUP, and you hear reorder tone.

2. Dial the pilot number of the ACD hunt group you want, or press pound (\#) to log out of all of your ACD hunt groups. You hear a confirmation tone, and the display shows AGENT LOGGED OUT OF <hunt group>.

   *If you were not logged in to the hunt group number that you entered,* the display shows NOT LOGGED INTO <hunt group>. Start over.

3. If off-hook, hang up.

**Restart ACD Idle Time Upon Login Flag**

The Restart ACD Idle Time Upon Login flag determines where an agent is placed in a longest idle queue when he or she logs back in to a hunt group. When enabled, the agent’s idle time is reset to zero whenever he or she logs in (i.e., that agent will be least likely to receive the next distributed call). If disabled, the agent’s idle time includes the time the agent was logged out of the hunt group (i.e., that agent will be most likely to receive the next distributed call). By default, this flag is disabled.

This flag is programmable for each individual hunt group.

**ACD Agent Wrap-Up Timer**

Each time an agent ends an ACD hunt group call, the endpoint goes into “wrap-up” mode, and the ACD Wrap-Up Duration timer is started. Until that timer expires, the agent will not receive another call through any ACD hunt group. However, the agent can receive non-ACD hunt group calls, direct ring-in calls, and transfers.

The range of the ACD Wrap-Up Duration timer is 1 to 65,535 seconds. The default value is 15 seconds. It is programmed individually for each ACD hunt group.

If an agent wants to terminate the wrap-up mode before the timer expires, he or she can use the following procedure.

**To terminate ACD agent wrap-up:**

While on- or off-hook (single line users must lift the handset), enter the ACD Wrap-Up Terminate feature code (329). The display shows ACD WRAP-UP TERMINATED. *If you were not logged in to an ACD hunt group,* the display shows CANNOT ACCESS RESERVED FEATURE.*)
Wrap-Up Mode for Holding ACD Calls

If enabled, the system flag called “Wrap-Up Mode For Holding ACD Calls” places an ACD agent’s endpoint in wrap-up mode when an ACD call is placed on Hold. However, the ACD Wrap-Up Duration timer is not activated.

This feature prevents the agent from receiving additional ACD hunt group calls after placing an ACD call on Hold while the endpoint is idle. The endpoint can still receive non-ACD calls, as usual. If the flag is disabled, the agent will be available to receive additional ACD calls as soon as an ACD call is placed on Hold. In the default state, this flag is disabled.

This flag applies to any type of Hold, including Individual, System, Transfer, and Conference-Wait Hold. If the call is terminated or if it is reverse transferred by another endpoint, the ACD agent’s endpoint will be made available to receive incoming ACD calls.

If the agent wants to terminate the Wrap-Up Mode while an ACD call is on Hold, the procedure described in the preceding section can be used.

ACD Agent ID Automatic Connect Flag

When the ACD Agent ID flag is enabled, and the agent is using a headset, ACD hunt group calls will automatically be connected following a short ring burst. This feature will not work if the agent is not using a headset.

When the ACD agent logs in or removes the endpoint from Do-Not-Disturb mode, the first call will ring until the ACD agent answers it. However, every time an ACD hunt group call is received after the first call, the agent will hear the ring burst from the endpoint and the call will be automatically connected.

This flag overrides the endpoint’s Transfer-to-Connect endpoint flag, which is disabled by default.

Group Call Pick-Up

When the Group Call Pick-Up feature is enabled, a call ringing in to a hunt group or one of its endpoints can be picked up at any other endpoint. Users can enter the Reverse Transfer feature code (4) and dial a hunt group’s pilot number to pick up a call that is ringing in to the hunt group’s pilot number or to any endpoint within that hunt group.

The system follows the priority list shown below to determine which call is picked up. The system follows the hunt group list (always beginning with the first endpoint on the list) to check each endpoint in the hunt group and then the overflow endpoint for one type of call at a time. If there is more than one call of the same type at the selected endpoint, the call that was received by the endpoint first is picked up. Holding calls and queue callbacks cannot be picked up.

1. Calls to the hunt group pilot number
2. CO calls
   a. Direct ring-in calls
   b. Transferred calls
   c. Recalls
   d. Camped-on calls
3. IC calls
   a. Direct ring-in calls
   b. Transferred calls
   c. Recalls
   d. Camped-on calls

Group Call Pick-up can retrieve calls only from endpoints that are currently logged in to the hunt group. You cannot use this feature to pick up calls from members who have logged out using the Hunt Group Remove feature code. Also, Group Call Pick-up cannot be used on ACD Hunt Groups that use Agent IDs. It can only be used on hunt groups that use lists of extensions. If the ACD Hunt Group flag is enabled, the Group Call Pick-up flag will be dimmed.

**To pick up a call that is ringing at a hunt group:**

1. Lift the handset, and/or enter the Reverse Transfer feature code (4). You hear a single progress tone.
2. Dial the pilot number of the hunt group where the call is ringing.

   **Inter-Tel endpoints:** If the system is programmed for direct connection of reverse transferred calls, you are immediately connected. If not, press the flashing CALL, individual trunk, or button.

   **Single line endpoints:** You are automatically connected.

### ACD Hunt Groups Span Nodes

The premium feature called Remote Automatic Call Distribution Hunt Groups allows ACD hunt groups to span nodes. Node-spanning ACD hunt groups can have either members or ACD Agent IDs.

- **Members:** ACD hunt group members may include off-node endpoints, off-node single lines, and Hunt Group Member extension lists.
- **ACD Agent IDs:** Unlike members, there are no off-node Agent IDs. Agent IDs are considered global throughout all nodes in which the Agent exists. This means that if you create Agent ID 100 on one node, you must also create Agent ID 100 on all other nodes that have an endpoint that a user may want to login to the hunt group using that ID.

The Remote ACD Hunt Groups premium feature must be included in the software license where the hunt group resides as well as any node that wants to have users login to the node-spanning hunt group. Because of this fact, several cases must be considered.

**Table 21** shows the different displays and results for logging in to and out of ACD hunt groups with the different combinations of the premium feature and the ACD Hunt Group enabled/disabled.

**Table 21. Displays and Results for Logging In and Out**

<table>
<thead>
<tr>
<th>Endpoint Node</th>
<th>Hunt Group Node</th>
<th>Login Display</th>
<th>Login Result</th>
<th>Login “All” Display</th>
<th>Login “All” Result</th>
</tr>
</thead>
<tbody>
<tr>
<td>Yes</td>
<td>Yes</td>
<td>AGENT LOGGED INTO HUNT GROUP XXXX</td>
<td>Agent logged into hunt group.</td>
<td>AGENT LOGGED INTO ALL ACDS</td>
<td>Agent logged into all local and remote groups.</td>
</tr>
</tbody>
</table>
Table 21. Displays and Results for Logging In and Out (Continued)

<table>
<thead>
<tr>
<th>Endpoint Node</th>
<th>Hunt Group Node</th>
<th>Login Display</th>
<th>Login Result</th>
<th>Login “All” Display</th>
<th>Login “All” Result</th>
</tr>
</thead>
<tbody>
<tr>
<td>Yes</td>
<td>No</td>
<td>CANNOT ACCESS RESERVED FEATURE</td>
<td>Agent not logged into group.</td>
<td>AGENT LOGGED INTO ALL ACDS</td>
<td>Agent logged into all local hunt groups, but not remote where feature disabled.</td>
</tr>
<tr>
<td>No</td>
<td>Yes</td>
<td>CANNOT ACCESS RESERVED FEATURE</td>
<td>Agent not logged into group.</td>
<td>AGENT LOGGED INTO ALL ACDS</td>
<td>Agent logged into all local, but no remote groups.</td>
</tr>
<tr>
<td>No</td>
<td>No</td>
<td>CANNOT ACCESS RESERVED FEATURE</td>
<td>Agent not logged into group.</td>
<td>AGENT LOGGED INTO ALL ACDS</td>
<td>Agent logged into all local, but no remote groups.</td>
</tr>
</tbody>
</table>

Programming ACD Hunt Groups

ACD hunt groups have options allowing the addition of endpoint off-node devices, single line off-node devices, and hunt group member extension lists containing local or off-node endpoints or both.

If the Remote Call Distribution Hunt Groups premium feature is not included in the node’s software license, the off-node options will not appear in the list when programming members for ACD hunt groups.

Operational Changes

The Station Monitor feature remains limited to a single node. Members who are logged in to a remote node or who are logged in to the supervisor’s node from a remote node cannot be monitored.

Hunt groups with remote members rely on network links among nodes. Node availability and software version compatibility affects hunt group log-in procedures as follows:

- Hunt group members attempting to log in to all hunt groups will be logged in to only those hunt groups on nodes that are reachable. The display of the user’s endpoint will only indicate log-in results for reachable nodes.
- Members logging out of all hunt groups will be logged out of groups on reachable nodes immediately and will automatically be logged out of groups on unreachable nodes as soon as the link to the node is restored. The endpoint displays MESSAGE PENDING in this case.
- When users attempt to log in to a remote hunt group on a node running system software that is not Inter-Tel 5000, the endpoint display shows NOT AN ACD HUNT GROUP.
- If users try to log into a hunt group on a node that is not Inter-Tel 5000, the display shows the message NOT AN AGENT IN THIS ACD GROUP.
The ACD Wrap-Up functions apply the following prioritization method for hunt groups with remote members:

- The first criterion is priority level. The hunt group with the highest priority gets the available agent first.
- In the event of a tie in priority, hunt groups located locally take priority over those located on remote nodes.
- In cases where ties in priority occur within the local node, the hunt group with the longest camped on call will receive the available agent.
- Given off-node ties in priority, the first hunt group to camp on will receive the available agent.

**Maintenance of Node-Spanning Hunt Groups**

- Deletion of an ACD agent from a hunt group is done from the node on which the hunt group resides or from a remote node.
  - If an agent is deleted on the node containing hunt group(s) of which he is a member, then the agent is removed from all hunt group(s) on that node and is logged out.
  - If the agent is deleted from a remote node, the agent is logged off from that node, but remains a member of any hunt group(s) located on other nodes.
- If the hunt group is deleted, all remotely logged-in endpoints will be logged out.
- If the extension number of an ACD Hunt Group is changed, any local or remote endpoints that are logged in will reflect the change. If links are down at the time of the change, remote nodes will not receive the change and an import/export process will be necessary to update the information on the affected nodes.
- Removing an Agent ID from a hunt group agent list logs out any endpoint, local or remote, that is logged in to that hunt group using that Agent ID.
- With the node-spanning features of ACD hunt groups, changing Agent IDs can be done from two possible locations: on the node where the group resides or on a node where the logged-in endpoint resides.
  - When changing an Agent ID on the node where the hunt group resides, the remotely logged-in endpoint using that Agent ID will be logged out even though the Agent ID still exists on the remote node. The hunt group will then contain the new Agent ID in its removed list, making the new Agent ID available for log-in use.
  - When changing an Agent ID on the node on which the logged-in endpoint resides (a node remote from the hunt group), the remote member will be logged out and the old Agent ID will no longer exist on that node. The hunt group on the original node will contain the old Agent ID in its removed list, because this ID still exists on that node.
- During a database restore on a node containing ACD hunt groups that use Agents IDs, all Agent IDs are placed into each hunt group’s removed list. All endpoints will be logged out during the restore.
- When the extension number of a logged-in endpoint is changed, the new extension will be logged in with the same Agent ID and will not be disconnected. This applies to endpoints logged in both remotely and locally.
Enabling the Send Camp-On Notifications to Members in DND Flag

The Send Camp-On Notifications to Members in DND flag is located under System\Devices and Feature Codes\Hunt Groups\Local\<Hunt Group>. This flag indicates whether or not camp-on burst tones are sent to hunt group members that are in DND or that are logged out. When sent to display endpoints, the display also indicates N CALLS WAITING FOR <HUNT GROUP>, where N is the number of calls that are currently camped on to the hunt group. This allows hunt group members to see the hunt group queue in real-time.

If this flag is enabled:

- UCD hunt group members that are in DND will receive Camp On notifications.
- ACD hunt group members that do NOT use agent IDs will receive Camp On notifications when the members are NOT logged in to the hunt group. This, in effect, is the same as being in DND from the hunt group perspective.
- ACD hunt group members that use agent IDs will receive Camp On notifications only when the agent is logged in to the hunt group and in DND. Agents also receive notifications while in wrap-up.

If the flag is disabled, which is the default setting, camp-on notifications are not sent to hunt group members.

Programming a Mini-DSS for Agent Wrap-Up

ACD agents can program a Mini-DSS button to light while they are in wrap-up mode. When the wrap-up timer has expired, the lamp turns off, and the agent is available for calls again. To use this feature, the agent’s endpoint must have an available user-programmable button.

To program a Mini-DSS for agent wrap-up at an agent endpoint:

1. Enter the Program Keys feature code (397 by default).
2. Press the button to program.
3. Enter the ACD Agent Wrap-Up Terminate feature code (329 by default). The button is now programmed for wrap-up indications.

Because the button is programmed with the ACD Agent Wrap-Up Terminate feature, agents can also press the button to be removed from wrap-up mode. When the agent terminates the wrap-up time, the lamp turns off.

Trunk Features

This section explains the trunk functions and programmable features.

NOTICE

Possible trunk misuse. While this system is designed to be reasonably secure against CO trunk misuse by outside callers, there is no implied warranty that it is not vulnerable to unauthorized intrusions and toll fraud. If the central office does not provide supervision it will not disconnect the call when one party hangs up; it is possible for a caller to remain connected to a CO trunk circuit. If this happens, and the caller begins dialing, the call could be placed through the system and would then be billed to the system’s owner. The system cannot check this type of call for toll restriction and may not register the call in SMDR. This problem could arise when a call is connected to an endpoint or when a call is in an unsupervised conference.
Caller ID, DNIS, and ANI

The system supports Caller ID, ANI, and DNIS to data about the source of the call.

**IMPORTANT**

The standard Advanced CO Interfaces feature is required to use the following features.

- **Caller ID**: Caller ID (CID) information provides the calling party’s phone number and/or name. With v2.2 and later, CID information received via an Integrated Service Digital Network (ISDN) line, analog line, or T1 circuit can be processed and sent out to an ISDN circuit. CID information can also be propagated between nodes.

- **Automatic Number Identification (ANI)**: ANI information identifies the caller’s phone number. The system receives a specified number of digits.

- ***ANI***: This is a form of ANI that does not have a specified number of digits. The system receives a star (*) before the ANI digits to signal the beginning of the caller’s phone number. Then there is another star after the digits to signal the end of the ANI information.

- **Dialed Number Identification Service (DNIS)**: DNIS identifies the number that was dialed to reach your location. The system receives a base number and a specified number of digits that identify the dialed number.

- ***DNIS***: This is a form of DNIS that does not have a specified number of digits. The system receives a star (*) before the DNIS digits to signal the beginning of the dialed number. Then there is another star after the digits to signal the end of the DNIS information.

- ***ANI*DNIS***: This is a two-stage address service that provides both the caller’s phone number and the dialed number. It combines the *ANI* and the *DNIS* features described above.

**Inter-Tel Endpoint Displays**

Whether call information appears on the Inter-Tel endpoint display and what information appears, is determined by endpoint flags in the database. The flags include:

- **Expanded CO Call Information on Displays**: This flag determines whether call information (trunk name or call information) is displayed at the endpoint. If it is enabled, the **Outside Call Party Information Has Priority** flag (described below) determines what is displayed. If it is disabled, the programmed trunk group username appears on the display. In the default state, this flag is **enabled**.

- **Outside Party Call Information Has Priority**: If the Expanded CO Call Information flag is enabled (as described above), this flag determines what information is displayed at the endpoint. If enabled, any call that is received on a trunk that provides outside call information (for example, Caller ID or ANI), is identified on the endpoint’s display with the call information. If disabled, the display shows the DID or DNIS information for the call (if available). In the default state, this flag is **enabled**.

- **Display Outside Name**: If this flag is enabled, the endpoint user can switch between the outside party name and number when connected to a CO call with outside party information. The endpoint user enters the Display Outside Party Name On/Off feature code (379). In addition, the enhanced ring-in displays provide the user with more information such as both Caller ID name and number if available, or tell the user if a Caller ID number is blocked or out-of-area. This is an endpoint flag that applies to Inter-Tel endpoints only. In the default state, this flag is **enabled**.
The system provides the user with advanced displays for direct ring-in calls if the endpoint has both of the following endpoint flags enabled: **Expanded CO Call Information on Displays** and **Outside Party Call Information Has Priority**.

A typical direct ring in display would look like the following: TRNK GRP 1 RINGING IN. However, with the enhanced displays, the system will attempt to display the name of the outside caller on line 1 and the number of the outside caller on line 2.

When a CO call rings into the system, the system uses the following criteria to determine the name that appears on the display endpoints. The criteria are listed from highest priority to lowest priority.

**Display Line 1**

The following criteria are used to display the name on the top line of the ring-in display. This assumes that all of the necessary endpoint flags are properly set and usernames are correctly filled in or left blank (at the programmer’s discretion) for Call Routing Tables and Trunk Group Names.

1. Outside Party Name provided by the Desktop Application, if available.
2. Outside Party Name provided by Caller ID, if enabled at the system-wide flag level.
3. Outside Party Name provided by System Speed Dial. If the collected number matches a number in a Speed Dial bin, the system uses the name of the corresponding Speed Dial bin.
4. Outside Party Number provided by the Desktop Application.
5. Outside Party Number provided by Caller ID, if enabled at the system-wide flag level.
6. Outside Party Number provided by ANI service, if enabled at the system-wide flag level.
7. Call Routing Table Name, if the name is not blank.
8. Trunk Group Name, if the name is not blank.
9. Default Trunk Group Name (TG XXXXX).

**Display Line 2**

The following criteria are used to display the number on the bottom line of the ring-in display. This assumes that all of the necessary flags are properly set.

1. Outside Party Number provided by the Desktop Application, if available.
2. Outside Party Number provided by Caller ID, if enabled at the system-wide flag level.
3. Outside Party Number provided by ANI service, if enabled at the system-wide flag level.
4. Number Absence Reason, such as Caller ID OUT-OF-AREA or PRIVATE (blocked) message.
5. RINGING IN display.

ANI and Caller ID information is also sent to the Voice Mail application to provide the caller’s phone number as part of the message envelope.

Caller ID is not transmitted to Single-Line Adapters (SLAs).
Outgoing-Access, Allowed-Answer, and Ring-In Assignments

Each trunk group has programmed lists of endpoints for outgoing-access, allowed-answer, and ring-in assignments for day and night modes.

**Outgoing-access** assignment for a particular trunk group or node trunk group permits the endpoint user to place calls using trunks in that trunk group. Each endpoint has a default outgoing access code programmed in the database. When the user presses the **OUTGOING** button, presses an idle Call button, or enters the Outgoing Call feature code (8), the system automatically selects an outgoing trunk.

Because the network does not allow users to directly access trunks on other nodes, each trunk group’s Outgoing Access can contain only endpoints on the local node. Users must use ARS to access trunks on other nodes.

- **Allowed-answer** access to a particular trunk group permits the endpoint user to answer incoming calls on the trunks in that trunk group, even if the endpoint does not have ring-in assignment for that trunk group. Endpoints cannot have allowed-answer assignment for trunk groups on other nodes. On Inter-Tel endpoints, the individual trunk button flashes (if one exists) to indicate the ringing call. If an endpoint is programmed with allowed-answer access only (no ring-in) for a trunk group, direct ring-in calls on that trunk group can be answered by entering the Automatic Trunk Answer feature code (350) or pressing the flashing individual trunk button.

- **Ring-in** assignment for a particular trunk group can be assigned to one or more endpoints, extension lists, or voice processing system applications; to a hunt group; to DISA; or to a call routing table. A trunk can ring in to an endpoint or voice processing system application on another node. The ring in destination(s) for the trunk group will receive direct ring-in calls on trunks in that trunk group. On Inter-Tel endpoints with ring in, the individual trunk or Call button flashes, and the display indicates a ringing call. Allowed answer for the trunk group is automatically assigned to an endpoint with ring in.

When Inter-Tel endpoints have outgoing-access or ring-in assignments, the associated individual trunk buttons show the status of their trunks.

Endpoints that do not appear on any of the lists cannot place or directly receive outside calls; they are limited to intercom calls, conferences, transferred calls, and retrieving calls on system hold. A call on system hold can only be picked up at the endpoint that placed it on hold or at an endpoint that has an individual trunk button and has allowed-answer and/or outgoing access for that trunk.

A private trunk group with one or more trunks can be established by programming outgoing-access, ring-in, and allowed-answer permission for the trunk group to only one endpoint.

**Answer Feature Code**

For calls that are ringing or holding at the endpoint, the user may enter the Answer feature code (351) or press the **ANSWER** button. When more than one call is ringing or holding, the following priority list determines which call is answered first:

- Ringing calls (ring ins, recalls, callbacks, or transfers) are answered in the order they were received.
- Calls on individual hold are answered in the order they were placed on hold.
Automatic Route Selection (ARS)

ARS is a money-saving feature that allows the system to be programmed to select the least expensive route for placing outgoing calls. It can be used for placing outgoing calls and transferring or forwarding calls to outside phone numbers. Endpoints can be restricted to using only ARS for placing outgoing calls. Also, because users do not have direct access to trunks on other nodes, ARS is the only way users can place calls using the other nodes’ trunks.

ARS is programmed using route groups and facility groups.

- **Route Groups**: A route group contains dialing patterns and facility groups.
  - The dialing patterns are used to determine the calls that will be routed through the route group. For example, the default dial pattern for Route Group 1 is N+ (any number of digits beginning with digit 2–9). If a number is dialed that begins with 1, it will not be routed through this route group.
  - Each route group has an ordered list of facility groups that contains lists of local trunk groups and/or nodes. There can be 100 facility groups in the system. The facility groups should be programmed so that the least-expensive route is checked and, if available, is selected first. If the least-expensive facility group is not available, the system checks the other groups in the list until it finds an available trunk.

- **Facility Groups**: A facility group contains trunk group/node lists and dial rules.
  - The trunks can include local trunk groups or nodes.
  - The dial rules tell the system what to dial. The system can have up to 32 dial rules, 26 of which are programmable. Each facility group can use up to 32 dial rules. For example, if the selected route group requires that the number contain “1” but no area code, the dial rules include the 1 and drop the area code. The modified phone number can contain up to 32 digits. (If SMDR is enabled, the modified phone number, not the digits dialed, will appear in the SMDR call record.) When programming ARS, you can use preset dial rules or create new dial rules that add up to 16 digits each.
  - The preprogrammed dial rules are:
    - **Dial Rule #1 – Echo Equal Access**: This non-programmable dial rule includes the equal access digits (10XXX or 101XXXX) in the number, if dialed.
    - **Dial Rule #2 – Echo Toll Field**: The toll field (1, 0, 01, or 011) is included in the number, if dialed. This dial rule is non-programmable.
    - **Dial Rule #3 – Echo 3 Digits After Toll Field**: With this non-programmable dial rule, the three digits after the toll field are included in the number. (These digits are usually the area code or country code.)
    - **Dial Rule #4 – Echo Local Address**: This non-programmable dial rule allows ARS to dial the phone number that was dialed by the user.
    - **Dial Rule #5 – Add Account Code**: This non-programmable dial rule causes the system to dial the account code that is associated with the call, when it is available. The account code can be entered using any of the account code types, including All Calls Following, as long as the system receives the account code before the end of dialing. (To use an optional account code, the code must be entered before the number is dialed.)
    - **Dial Rule #6 – Add Extension Number**: This non-programmable dial rule tells the system to include the extension number of the endpoint or voice processing system application that is being used to place the call.
    - **Dial Rule #7 – Add #**: This dial rule adds a pound (#) to the number dialed. This dial rule can be programmed.
When ARS is selected, the user dials the number (including the area code, if needed), and the system does the following:

1. **Checks the dialed number and matches the dialing pattern to a route group:** The system checks the route groups in numerical order and selects the first group that applies to the dialing pattern of the number that was dialed.

   ![NOTE]
   
   If the Emergency Call feature is programmed to use ARS, the emergency calls will always go through Route Group 1, regardless of route group programming.

2. **Selects a facility group:** If all of the trunks are busy, and the endpoint is enabled for ARS camp on, the call will camp on to the facility group until a trunk is available.

3. **Checks for toll restrictions and outgoing access:** Before the number is modified by the facility group’s dial rules, the system checks the endpoint toll restrictions and outgoing access to determine whether the call is allowed. If allowed, the system continues to the next step. If not allowed, the system sends reorder tones, and the call is not placed. All calls placed using ARS are toll restricted, regardless of whether the selected trunk is subject to toll restriction. However, Emergency Call feature calls are never toll-restricted.

4. **Adds or deletes digits according to the facility group chosen:** Each facility group has a programmed set of dial rules that tells the system what to dial. For example, if the selected route group requires that the number contain “1” but no area code, the dial rules include the 1 and drop the area code.

5. **Dials the modified phone number:** If the number is allowed, the system seizes an idle trunk in one of the selected trunk groups, waits for the Dialing Wait After Connect timer to expire, and then dials the number.

When ARS is used, the user hears dial tone when the feature code is entered (manually or automatically using a Call button, the ARS button, or the OUTGOING button). The user then dials the number and hears silence until the trunk is seized and dialing is completed. (The user will not hear the digits being dialed.) The call appears under an available Call or trunk button.

**Figure 9. ARS Example**

ARS Example

[Diagram showing ARS process]

ARS does the following:

- Checks the number and finds dialing pattern 602+ in Route Group 2.
- Selects a facility group with an available trunk.
- Checks toll restriction and outgoing access.
- Uses dial rules to modify the number:
  - Echo 3 Digits After Toll
  - Echo Local Address
  - Add Account Code
- Dials the modified number.
Basic Rate Module (BRM-S)

Each Basic Rate Module (BRM-S) provides two BRI S/T-interface ports for connecting BRI trunks to the system. The BRM-S does not support video conferencing or connection of BRI endpoints to the system.

Each ISDN BRI port provides two trunks/circuits/bearer-channels for a total of four BRI trunks possible on each BRM-S module in the Inter-Tel 5000 Base Server.

Basic Rate Interface (BRI) capability is supported only in Inter-Tel Europe installations.

Toll Restriction

The following features provide toll restriction on the system and are described in further detail in this section.

- **Trunk Group Toll Restriction**: Each trunk group can be designated as “subject to toll restriction” or “not subject to toll restriction” in DB Programming. If subject to toll restriction, endpoint class of service is checked when the trunk group is selected for placing an outgoing call. The exceptions to trunk group toll restriction are as follows:
  - **Emergency Call Feature**: When the Emergency Call feature code is entered, the designated trunk access code is used to seize a trunk and the programmed emergency number is dialed regardless of any endpoint or trunk group toll restrictions.
  - **Automatic Route Selection (ARS)**: Calls placed using ARS (except Emergency Call feature calls) are always subject to toll restriction, regardless of the selected trunk group’s toll restriction programming. Note that only the endpoint COS is checked; the trunk group COS is not.
  - **Network ARS Calls**: When determining toll restriction for an ARS outgoing call, the network checks the endpoint toll restriction based on the database information on the node on which the endpoint resides, not the node that contains the trunk which the system uses to place the call. The system does not check the trunk COS for ARS calls.

- **Exemption from ARS-Only Restriction**: Each trunk group may be designated as “exempt from automatic route selection (ARS) only.” If exempt, endpoint users with the ARS-Only class of service can select the trunk group directly.

- **Absorbed Digits**: Trunk groups that are subject to toll restriction can be programmed to “absorb” (ignore) the first digit(s) dialed. This allows the system to handle the dialed digits just as they would be by the local telephone company or PBX to which the system is connected.

- **Endpoint and Trunk Group Classes of Service**: Each endpoint and trunk group can be assigned one or more classes of service (COS) to restrict or allow certain digit patterns when an outside call is dialed. Trunk group COS is not checked if ARS is used.
Class of Service

Each endpoint and each trunk group that is subject to toll restriction is assigned a toll restriction class of service (COS) that restricts dialing patterns on outside calls.

COS is programmed individually for endpoints, voice processing system applications, and trunk groups. Separate COS designations are available for day and night modes. An endpoint or trunk group can be completely unrestricted or can have a combination of COSs.

System Speed Dial numbers can be programmed to bypass COS restrictions on a system-wide basis. If the option is not enabled, all System Speed Dial numbers are subject to toll restriction.

Because a condition may exist where a critical or life-threatening situation needs to be reported, using the Emergency Call feature code will override all toll restrictions and trunk access programming.

Note that if the trunk group is not subject to toll restriction, neither the trunk group nor endpoint COS is checked, unless the call is placed using ARS. ARS calls are subject to endpoint toll restriction only.

The default US and European values for COS appear in the following tables.

Table 22. Class of Service Default US Values

<table>
<thead>
<tr>
<th>COS</th>
<th>Name</th>
<th>Restriction</th>
</tr>
</thead>
<tbody>
<tr>
<td>01</td>
<td>ARS Only</td>
<td>This is an endpoint class of service only. It is not used for trunk groups. Also, the Automatic Route Selection (ARS) feature, which is standard on the Inter-Tel 5000 platform, is required. Calls can only be placed using the ARS feature. The user will hear reorder tones when attempting to place a call using any other method. A restricted user can still select individual trunks if the trunks are designated as “exempt from ARS Only,” as described on page 142, or if they were transferred, were placed on hold, or are recalling or ringing. Trunk restriction determines which trunks in the ARS route group can be selected by the endpoint or application.</td>
</tr>
<tr>
<td>02</td>
<td>Deny Area/Office</td>
<td>This restriction is divided into eight user groups to allow the use of varying area/office code restriction tables. This is useful for reducing restrictions for some of the endpoints, applications, or trunk groups while increasing restrictions for others. Each endpoint, application, and trunk group is assigned a day mode and a night mode user group in DB Programming. Within each user group, area codes can be designated as restricted, allowed, or extended. Restricting an area code prevents users from placing calls to that area code. Allowing an area code allows all office codes within that area code. Designating an area code as extended allows the programmer to determine which office codes (up to 800) are allowed or restricted within that area code.</td>
</tr>
<tr>
<td>03</td>
<td>Deny Operator</td>
<td>Calls to operator numbers are restricted.</td>
</tr>
<tr>
<td>04</td>
<td>Deny Toll Access</td>
<td>Calls to numbers containing a toll string (defaults to 1) are restricted.</td>
</tr>
<tr>
<td>05</td>
<td>Deny International</td>
<td>Calls to international numbers are restricted.</td>
</tr>
<tr>
<td>06</td>
<td>Deny Equal Access</td>
<td>Calls that begin with equal access digits (10XXX or 101XXXX) are restricted.</td>
</tr>
</tbody>
</table>
Table 22. Class of Service Default US Values (Continued)

<table>
<thead>
<tr>
<th>COS</th>
<th>Name</th>
<th>Restriction</th>
</tr>
</thead>
<tbody>
<tr>
<td>07</td>
<td>Deny Local Calls</td>
<td>Calls to local numbers are restricted.</td>
</tr>
<tr>
<td>08</td>
<td>Denied Numbers</td>
<td>Calls to programmed “denied” numbers (defaults to 1900NXXXXXXX+ and 976XXXX+) are restricted.</td>
</tr>
<tr>
<td>09</td>
<td>Allowed Numbers</td>
<td>Calls to programmed “allowed” numbers (defaults to 1800NXXXXXXX+) are allowed, even if number also matches a dial pattern in a restricted class of service that is assigned to the endpoint, application, or trunk group being used.</td>
</tr>
</tbody>
</table>

The first nine COS designations (01–09) have default values. COS designations (10–16) do not have default values and are blank. All of the classes of service are programmable. Each may be specified as an “allowed” or “denied” class of service, dialing patterns can be programmed for each, and each can be assigned to endpoints and trunk groups as needed.

Table 23. Class of Service European Default Values

<table>
<thead>
<tr>
<th>COS</th>
<th>Name</th>
<th>Restriction</th>
</tr>
</thead>
<tbody>
<tr>
<td>01</td>
<td>ARS Only</td>
<td>This is an endpoint class of service only. It is not used for trunk groups. Also, the Automatic Route Selection (ARS) feature, which is standard on the Inter-Tel 5000 platform, is required. Calls can only be placed using the ARS feature. The user will hear reorder tones when attempting to place a call using any other method. A restricted user can still select individual trunks if the trunks are designated as “exempt from ARS Only,” as described on page 142, or if they were transferred, were placed on hold, or are recalling or ringing. Trunk restriction determines which trunks in the ARS route group can be selected by the endpoint or application.</td>
</tr>
<tr>
<td>02</td>
<td>Deny Operator</td>
<td>Calls to operator numbers are restricted.</td>
</tr>
<tr>
<td>03</td>
<td>Deny Toll Access</td>
<td>Calls to numbers containing a toll string (TN+, TE, 010+, and T1+) are restricted.</td>
</tr>
<tr>
<td>04</td>
<td>Deny International</td>
<td>Calls to international numbers (I+) are restricted.</td>
</tr>
<tr>
<td>05</td>
<td>Deny Local Calls</td>
<td>Calls to local numbers (N+) are restricted.</td>
</tr>
<tr>
<td>06</td>
<td>Denied Numbers</td>
<td>Calls to programmed “denied” numbers (0891+ and 0898+) are restricted.</td>
</tr>
<tr>
<td>07</td>
<td>Allowed Numbers</td>
<td>Calls to programmed “allowed” numbers (0345+, 0500+, 0645+, and 0800+) are allowed, even if number also matches a dial pattern in a restricted class of service that is assigned to the endpoint, application, or trunk group being used.</td>
</tr>
</tbody>
</table>

Calls are restricted only if the dialed patterns match the denied pattern exactly. The first nine COS designations (01–07) have default values. COS designations (10–16) do not have default values and are blank. All of the classes of service are programmable. Each may be specified as an “allowed” or “denied” class of service, dialing patterns can be programmed for each, and each can be assigned to endpoints and trunk groups as needed.
Emergency Call

NOTICE

It is the responsibility of the organization and person(s) performing the installation and maintenance of Inter-Tel Advanced Communications Platforms to know and comply with all regulations required for ensuring Emergency Outgoing Access at the location of both the main system and any remote communication endpoints. Ask your Inter-Tel equipment provider or Inter-Tel-certified technician for details about Inter-Tel systems’ Emergency Call capability. Remote IP and SIP endpoints may require gateway access to nearby emergency responders. Emergency Call phone numbers include:

- **911**, the default for Inter-Tel systems located in the United States (US).
- **999**, the default for Inter-Tel systems located in the European market and used primarily in the United Kingdom (UK).
- If applicable, **112**, an emergency number used widely in Europe outside of the UK.

Any emergency number, such as for a police or fire station, that is appropriate for the location of the main system and/or remote endpoints.

The Inter-Tel 5000 allows immediate access to local emergency facilities when an endpoint user enters the Emergency Call feature code. The dialing pattern defaults to 911 in US systems and to 999 on systems in the European market. When activated, the Emergency Call feature selects a trunk or routes the call based on endpoint programming. When activated, the Emergency Call feature overrides all toll restrictions and trunk access programming.

Emergency Extensions can be programmed for each endpoint, and Day/Night Emergency Outgoing Access Lists are used to validate the extension making the emergency call. When a user dials the Emergency Call phone number at intercom or CO dial tone, emergency outgoing access is granted based on system programming and what the user dialed, as described in the following scenarios:

- **If the user dials the Emergency Call feature code from an endpoint with a trunk or trunk group programmed as the Emergency Extension**, the Emergency Call feature routes the call based on the endpoint’s Emergency Extension. The trunk or trunk group does **not** validate the endpoint originating the emergency call against the trunk group’s Emergency Day/Night Outgoing Access List. An idle trunk is seized and the system automatically dials Emergency Number 1.

- **If the user dials the Emergency Call feature code from an endpoint with ARS programmed as the Emergency Extension:**
  
  a. The Emergency Call feature routes the call based on the endpoint’s Emergency Extension.
  
  b. The call is routed to Route Group 1.
  
  c. The call tries the first facility group in Route Group 1.
  
  d. The trunk group or node trunk group validates the endpoint originating the emergency call against the group’s Emergency Day/Night Outgoing Access List.
e. If the call is denied, it tries the next member in the facility group. If each member
denies the call, the call tries the next facility group in Route Group 1’s list.

If everything is denied due to Emergency Outgoing Access, the call is routed once
again to Route Group 1. The system then tries to route the call on the first facility
group in Route Group 1. The trunk group or node trunk group does not validate
the endpoint originating the emergency call against the group’s Emergency Day/
Night Outgoing Access List. If no trunks are available, the system tries to route
the call on the next member in the facility group. If all of the members are
unavailable, the call will Camp On to Route Group 1 until a trunk is available.

• If the user accesses a trunk or trunk group and dials any Emergency Number (1–10),
the trunk or trunk group does not validate the endpoint originating the emergency call
against the trunk group’s Emergency Day/Night Outgoing Access List. An idle trunk is
seized, and the system automatically dials the emergency number.

• If the user accesses ARS and dials any Emergency Number (1–10):
  a. The call is routed to Route Group 1.
  b. The call tries the first facility group in Route Group 1.
  c. The trunk group or node trunk group validates the endpoint originating the emer-
gency call against the group’s Emergency Day/Night Outgoing Access List.
  d. If the call is denied, the call tries the next member in the facility group. If each
     member denies the call, the call tries the next facility group in Route Group 1’s
     list.

If everything is denied due to Emergency Outgoing Access, the call is routed once again
to Route Group 1. The system then tries to route the call to the first facility group in Route
Group 1. The trunk group or node trunk group does not validate the endpoint originating the emerg-
ency call against the group’s Emergency Day/Night Outgoing Access List. If no trunks are
available, the system tries the next member in the facility group. If all of the members are
unavailable, the call camps on to Route Group 1 until a trunk is available.

IMPORTANT

Inter-Tel recommends that only local trunks be installed and used for emer-
gency trunk access and that only local trunks be programmed in Route
Group 1. Programming a node in Route Group 1 may cause the system to
access a trunk on a separate node when ARS is used for emergency
access. If this occurs, Emergency Outgoing Access is no longer validated.

The system allows the Dialing Wait After Connect timer to expire and then dials the digit
string programmed in the database as the Emergency Number.
At the time the call is processed, a minor alarm is generated by the system and sent to all administrator endpoints. Also, if the Message Print option is enabled, the alarm message is sent to the designated output port. See Figure 10.

**Figure 10. Emergency Call Feature Flow Diagram**

![Emergency Call Feature Flow Diagram](image)

### Emergency Outgoing Access Lists

The Inter-Tel 5000 supports two outgoing access lists called *Emergency Day Outgoing Access* and *Emergency Night Outgoing Access*. These lists, which are similar to the Day/Night Outgoing Access lists, are programmable for CO Trunk Groups and Node IP Connection Groups. These lists default to the Auto Extension List PP051: Auto: All Endpoints.

### Ten Emergency Numbers

The system can store up to 10 emergency numbers that are dialed when the Emergency Call feature is used.

When the Emergency Call feature code is dialed, the system selects a trunk from Trunk Group 1 and automatically dials the number programmed in the database as Emergency Number 1. For systems operating in the US, the default Emergency Number 1 is 911. For systems operating in the European market, the default Emergency Number 1 is 999.

If a user accesses a trunk or ARS and dials Emergency Numbers 1–10, the dialed number will automatically override all toll restrictions and trunk access programming.

If the system is installed in an area where emergency responder services such as 911 or 999 are not available, Inter-Tel recommends substituting the phone number for the local police or fire department or the telephone company operator.

This feature can be made inoperative by removing all trunk access and/or by removing the dialed digit string. If Emergency Call is not operational, the system presents a warning message that the feature has been disabled. The programmer must acknowledge or change this condition before system operation can continue.

Like other feature codes, the Emergency Call feature code can be changed from 911 or 999 to a different code, if necessary.

An Emergency Number is blocked when no trunks or Emergency Numbers are programmed in the database or when all trunks in a trunk group are busy.

If the Emergency Call feature is programmed to use ARS, Route Group 1 for local calls will be used even if its dial patterns are reprogrammed.
Emergency calls, by default, use the first local trunk group and will not be sent using node trunk groups on other nodes. However, when ARS is used to place an emergency call, Route Group 1 is used even if it contains nodes. This means that the network can access a trunk on a node other than the user’s node if the user accesses ARS and dials the Emergency Number. When a user places an emergency call, every administrator in the network receives the emergency alarm, Alarm #11.

**Important**

Local trunks must be installed and used for Emergency Number trunk access. Nodes should **not** be used in Route Group 1.

**Calling Party Number**

Each endpoint can be programmed to send an identifying number when a call is placed. The Calling Party Number field is located in Devices and Feature Codes/Endpoints/Individual Endpoint programming. This information is required by government regulation for emergency calls in areas. Up to 48 digits can be programmed in the Calling Party Number field. However, check with your service provider to determine their specific requirements for this field. This number will be sent in the ISDN setup message in the Calling Party Number Information Element. In addition, the system will also send the extension number of the endpoint in the Calling Party Number Subaddress Information Element.

**Important**

No default number exists for this field. It is up to the programmer to supply the correct Emergency Calling Party Number for each endpoint.

If an Off Premises Extension is used for dialing and Emergency Number such as 911 in the US or 999 in the European market, the emergency responder operators will see Calling Party Number or the address of the location of the system, not the off-premises address. Off-premises personnel should be prepared to give the correct address and other pertinent information if it is not programmed as the Calling Party Number.

**Figure 11. Example of a Remote Site Without an MGCP/SIP Gateway**

- **Remote Site**
  - Located in Orange, California USA
  - 714-283-1600

- **IP Endpoint**

- **Caller**
  - 911

- **IP NETWORK**

- **PSTN**
  - Inter-Tel 5000
    - Located in Phoenix, Arizona USA
    - 480-961-9000

- **911 Operator**
  - Phoenix, Arizona USA
  - 480-961-9000

- **Caller ID Display:**
  - 714-283-1600
The MGCP/SIP gateway is intended to connect to the PSTN. Connecting the MGCP gateway to single line ports on other telephone systems is currently not supported.

When configured with an MGCP/SIP Gateway, the system can be programmed so that when an Emergency Number such as 911 or 999 is dialed from an IP endpoint, the associated calling name and number, not the system, is sent to the emergency responder operators.

**WARNING**

**Possible Delay in Local Emergency Response to Remote Sites.**

IP and SIP endpoint users should be alerted to the following hazardous situations:

- If an Emergency Call phone number is dialed from an IP or SIP endpoint located at a remote site that is not equipped with a correctly configured gateway, the call will be placed from the location where the system Inter-Tel 5000 Base Server is installed rather than from the location where the emergency call is made.

- In this situation, emergency responders may be dispatched to the wrong location. To minimize the risk of remote site users misdirecting emergency responders, Inter-Tel recommends regular testing of MGCP/SIP gateway trunk(s) for dial tone.

- If uninterruptible power supply (UPS) protection has not been installed as part of the Inter-Tel 5000 system, IP and SIP endpoints will not operate when electrical power fails either at remote sites or at the main system location.

- To place calls during a power failure in this situation, IP and SIP endpoint users can only use a single line endpoint connected to one of the power failure bypass circuits built-in to the system Inter-Tel 5000 Base Server. If an endpoint connected to a power failure bypass circuit is not available, users should make emergency calls from a local phone not connected to the system.
Day and Night Modes

Separate lists appear in the database for Class of Service, Outgoing Access, Allowed Answer, and Ring In assignments for day and night modes. When an administrator enters the Night Ring feature code, the system uses the night mode lists. When the system is in day mode, the day lists are used. Active calls are not affected when the mode is changed.

System administrators can place any or all remote nodes into night mode or day mode. The default feature code for Enable Network Night is 9861. The default feature code for Disable Network Night is 9862.

The Night Ring On/Off feature code (9860) affects only the node on which the administrator resides. For a complete explanation of system administrator features, see “System Administrator Procedures” on page 27.

Direct Inward System Access (DISA)

DISA is a programmable feature that allows an outside party to dial into the system from an external DTMF endpoint and then dial extension numbers, hunt group pilot numbers, and off-node device extensions. DISA callers do not have access to outgoing trunks or page zones.

Any of the trunk groups can be programmed to receive incoming DISA calls in day and/or night mode. When not in use for DISA, the trunk group can be used for placing outgoing calls by endpoints with outgoing access permission.

Due to the natural characteristics of the trunk, the volume level of DTMF tones transmitted over the trunk may be substantially reduced before reaching the system. This natural degradation in tone volume may adversely affect the reliability of the DISA feature. Other factors which can affect DISA performance are trunk noise and the quality and strength of the DTMF tones generated by the off-premises endpoint itself. If the system cannot recognize a DTMF digit, the call is automatically sent to the primary attendant.

When a DISA user calls an extension number, the call rings as a direct ring-in call, even if the called endpoint is busy or in Do-Not-Disturb. The DISA caller hears music or ringing (determined by a system-wide option in DB Programming) until the Transfer Available or Transfer Busy timer expires. Then, if the call is not answered, it recalls the called endpoint’s attendant. If the called endpoint is forwarded, the call follows the programmed forwarding path.

When a hunt group pilot number receives a call through DISA, the call rings or circulates according to how the hunt group is programmed (i.e., linear or distributed).

If a caller dials a valid hunt group pilot number that has no members assigned to it, the call automatically rings at the primary attendant endpoint until the Abandoned Call timer expires. If the call is not answered before the Abandoned Call timer expires, the call is disconnected.

Security Codes

DISA trunks can be assigned security codes (up to eight digits in length) that are required for access to the system. The installer can program separate codes for each DISA trunk group to be used during day and/or night modes. To prevent unauthorized access to the outgoing trunks, all trunk groups using DISA should have a security code.

The database programmer can determine the number of times that a caller may unsuccessfully attempt to enter a security code and/or dial an extension number. If the user does not succeed within the determined number of attempts, the call will be disconnected (if the security code is invalid) or transferred to the primary attendant (if the extension number is invalid).
Using DISA

If DTMF decoders are unavailable when a DISA call is received, the incoming DISA call is automatically sent to the primary attendant.

To use DISA:

1. From a DTMF endpoint, dial the phone number of the DISA trunk. When the call is answered by the system, you hear system intercom dial tone.

   *If you hear a single progress tone,* enter the appropriate (day or night) DISA security code, followed by pound (#).

2. Dial an extension number or a hunt group pilot number. (To call the primary attendant, dial 0.) You hear music or ringing until the call is answered.

**Figure 13. DISA Example**

Inter-Tel Endpoint Features

The following sections describe how to use Inter-Tel endpoint features.

**Volume Controls**

On Inter-Tel endpoints, volume is controlled in the following ways:

- On Model 8600, 8620/8622, and 8660 endpoints, illustrations of speakers ( and ) represent higher and lower volume levels. Press the end of the button to lower the volume and the end of the button to raise the volume.

- On the Executive, Standard, and Axxess Basic endpoints, an illustration on the button (that looks like this ) represents higher and lower volume levels. Press the lower end in place of the down arrow and the higher end in place of the up arrow.

- On the Professional, Associate, and Eclipse Basic endpoints, arrows on the key represents higher and lower volume levels.

- The other Inter-Tel endpoints have separate VOL UP and VOL DOWN keys.

The volume adjustment button on the digital endpoints also can be used for scrolling through displays.
To change volumes on an Inter-Tel endpoint:

Press the appropriate button to raise or lower the volume, using one of the procedures listed below. A confirmation tone will alert you when you have reached the highest or lowest possible volume. On display endpoints, the display shows VOLUME LEVEL X. The “X” represents the level and changes as the volume is raised or lowered.

- **To change handset intercom voice volume**: Adjust the volume while on an off-hook (handset) intercom call.
- **To change handsfree intercom voice and paging volume**: Adjust the volume while on an on-hook (speakerphone) intercom call.
- **To change handset outside call voice volume**: Adjust the volume while on an off-hook (handset) outside call.
- **To change handsfree outside call voice volume**: Adjust the volume while on an on-hook (speakerphone) outside call.
- **To change headset intercom voice volume**: (Model 8620 and 8660 endpoints only) Adjust the volume while using the headset and on an intercom call.
- **To change background music volume**: Adjust the volume while listening to music through the endpoint speaker. The selected background music volume level is automatically saved.
- **To change alerting tone (ring tone) volume**: Adjust the volume while the endpoint is ringing or idle. OR, while on-hook, press both edges of the Volume button to hear the currently programmed volume. Then adjust the volume.
- **To change handset progress tone volume**: Press the appropriate arrow while listening to intercom dial tone through the handset.
- **To change speakerphone progress tone volume**: Press the appropriate arrow while listening to intercom dial tone through the speakerphone.
- **To save the setting on a digital endpoint**: Press both edges of the Volume button at the same time. The display shows VOLUME SAVED. *If this step is not performed*, the endpoint will return to the previous volume level when the call or tone ends, except for the ringing alert tone or background music volume, which is automatically saved at the most recent volume level.

To return an Inter-Tel endpoint to default volume levels:

Entering this feature code also cancels Do-Not-Disturb, manual call forwarding, background music, ring intercom always, all calls following account code, headset mode, and queue requests; and restores handsfree mode, pages, hunt group calls, and system forwarding.

While on- or off-hook, enter the Default Endpoint feature code (394). *If off-hook*, hang up. The display shows ENDPOINT DEFAULTED.
Selectable Ring Tone

The tone of the endpoint ring signals (all tones except call waiting) can be changed by the user to create distinctive ringing. If endpoints are placed close together, changing the tone makes each endpoint’s ring easier to recognize. On all Inter-Tel endpoints, the tone is changed by entering a feature code and a code, as described below.

There are ten ring tone options (0–9). Selection 0 will turn off the ring signals.

**To change endpoint ring tone:**

1. While on-hook, enter the Program Ring Tone feature code (398). The display shows SELECT RING TYPE (0–9) OR SCROLL.

2. **To select a specific ring tone:** Press a number in the range 1–9 for the ring tone you want. The display shows RING TYPE X SELECTED.

   **To scroll through the tones:** Press the low end of the Volume button or the PREVIOUS button to scroll backward through the tones. Press the high end of the Volume button or the NEXT button to scroll forward. (You cannot scroll to selection 0, you must enter it as described below.) The display shows RING TYPE X SELECTED.

   **To turn off ringing:** Enter 0 or press the RINGER OFF menu button. The display shows RING TYPE OFF SELECTED.

3. Lift and replace the handset or press the pound button ( # ), both ends of the Volume button, or the ACCEPT button to exit. The selected tone will be heard when the endpoint rings (unless ringing was disabled). The display shows RING TYPE X SAVED.

Default Endpoint Identification on Display

When a display endpoint is idle, the extension number, username, time, and date are displayed. While on a call, in Do-Not-Disturb, or unconditionally forwarded, other information appears on the display. To redisplay the default ID, use the following procedure.

**To display the date, time, username, and extension number:**

On a Model 8600, 8620, 8660, or 8662 endpoint, press the Special button and enter the Display Time/Date feature code (300). The date, time, username, extension number, and endpoint status are displayed for 5 seconds. If the endpoint is operating in SIP mode, entering code 300 will display the IP address of the endpoint.

“All Transient Displays” Endpoint Flag

The “All Transient Displays” endpoint flag, when enabled, allows all transient call handling displays to appear on the endpoint display. If it is disabled, the CALL TRANSFERRED TO XXXX and MESSAGE LEFT FOR XXXX transient displays does not appear.

Each transient display appears for 5 seconds before another display can take its place. Disabling the flag allows high-traffic stations such as attendants to change displays more quickly after a message is left by the endpoint user or after a call is transferred.

This endpoint option cannot be programmed by the user. It is a database-programmable flag.
“DKTS Alternate Transient Display Timer” Endpoint Flag and Timer

As an alternative to (or in addition to) disabling the call transfer and message transient displays, as previously described, the programmer can use a combination of an endpoint flag and a timer to control **all** of the endpoints’ transient displays (not just the call transfer and message displays). The “DKTS Alternate Transient Display Timer” endpoint flag, if enabled, allows all transient displays to be controlled by the “DKTS Alternate Transient Display” timer. The “DKTS Alternate Transient Display” timer can be set to 1–255 tenths of a second (0.1–25.5 seconds). The default value is 10 tenths of a second (1 second). This timer affects only those Inter-Tel endpoints with the “DKTS Alternate Transient Display Timer” flag enabled.

“Transient Call Indication On Call Answer” Endpoint Flag

This programmable endpoint flag determines whether the endpoint user sees a call display when answering a call by pressing a secondary extension button or by reverse transferring. If enabled, the display shows if the call was ringing, recalling, transferred, or holding at the other endpoint.

User-Programmable Feature Buttons

This feature does not apply to single line endpoints.

Individual Inter-Tel endpoint buttons can be designated “user-programmable” in the keymaps in DB Programming. The installer can program the default feature codes, and the endpoint user can reprogram the feature codes as appropriate. Default endpoint feature codes are shown on page 395. Up to 35 user-programmable endpoint buttons can be programmed to enter any of the general feature codes.

Inter-Tel display endpoints have a special \[FWD\] key if the buttons are in the default state. However, it can be programmed to use any of the Forward feature codes. When the \[FWD\] button is pressed, the display shows a menu of call forwarding options. The user must then select the appropriate option by pressing the corresponding button next to the display. If the \[FWD\] button is programmed to use any other Forward feature code (355–358), the forwarding options will not be displayed when the \[FWD\] button is pressed. Any Inter-Tel endpoint user can make any user-programmable button a \[FWD\] button that uses any one of the forwarding feature codes (355–358).

Secondary extension buttons, described on page 93, can be assigned to user-programmable buttons during keymap programming. If so, the endpoint user can determine the primary endpoint that is associated with the secondary extension button by using the feature button programming procedures given on page 155.
To display the current button entries:

1. While on-hook, enter the Review Keys code (396). The display shows PRESS THE KEY TO REVIEW.

2. Press the feature or Speed Dial button(s) to be displayed. The display shows the feature, username, or speed dial location name and KEY VALUE: \(<feature code, extension number, or location code>\). For a secondary extension button, the display shows SECONDARY: \(<endpoint>\).

To return the buttons to default values:

While on- or off-hook, enter the Feature Key Default feature code (395). If off-hook, hang up. The display shows FEATURE KEYS DEFAULTED.

To program a feature code in a user-programmable button or assign a primary endpoint to a secondary extension button:

1. While on-hook, enter the Program Keys feature code (397). Display endpoints show PRESS THE KEY TO PROGRAM.

2. Press the feature or secondary extension button you want. The display shows the current feature, endpoint, or Speed Dial location name and KEY VALUE: \(<feature, extension, or location code>\).

3. Enter the feature code or extension number to be stored under that button. (If programming the button, only Call Forwarding feature codes 355–358 will be accepted.) You hear a single progress tone when the programming is completed. Display endpoints show the newly programmed feature.

   If you attempt to program a nonprogrammable button or enter an invalid code, you hear repeating reorder tones, and the feature code assigned to the button will remain unchanged. Display endpoints show NON-PROGRAMMABLE KEY or FEATURE CODE INVALID. If you do not enter a feature code before the Long Interdigit timer expires, the feature code assigned to the button is not changed.

4. To program other buttons, repeat the procedure.

To create a Speed Dial location in a user-programmable button:

1. While on-hook, enter the Program Keys feature code (397). Display endpoints show PRESS THE KEY TO PROGRAM.

   Speed-dial buttons are programmed using this procedure. Entering an extension number, using the above instructions, will create a DSS button for that extension number, not a Speed Dial button, even if the button was previously programmed as a Speed Dial button.

2. Press the feature button you want. The display shows the current feature, user, or speed dial location name and KEY VALUE: \(<feature or name>\).

3. Enter the Station Speed Dial feature code (382) or the System Speed Dial feature code (381). The display shows SELECT STN SPEED or SELECT SYS SPEED.

4. Enter the appropriate Speed Dial bin number (0–9 for endpoint, 000–999 for system). The display shows the current contents of that bin.

5. To program a new number in a Station Speed Dial bin, follow the instructions given on page 222. Administrators can use the procedures on page 50 to program System Speed Dial locations.
Standard/Alternate Keymap Switching

In DB Programming, each Inter-Tel endpoint is assigned a standard keymap and can also have an alternate keymap. The user can switch keymaps by entering a feature code.

Having two keymaps provides the user with access to more features or trunks. In most situations, the user would have one button programmed with the Switch Keymap feature code (399) in each keymap so that it would be easy to change the keymap. (If the user-programmed button has a lamp, the lamp is lit when the alternate keymap is active.)

When keymaps are switched, any user-programmable buttons in the maps retain their user-programmed values; they are not returned to default during the switch. However, if the keymaps do not have the same user-programmable buttons in the same positions, the location of individual buttons may change when the keymaps change.

If a user switches keymaps during an outside call, the call will appear on the appropriate trunk or Call button for that keymap. The button assignment may not be the same for both keymaps. For example, it could appear under a Call button on one map and an individual trunk button on the other.

If a button is not available in the currently active keymap when a call rings in, the user will still see the correct display, hear ringing, and can answer as usual.

To switch between keymaps when the endpoint is idle:

While on-hook, enter the Switch Keymap feature code (399). The display shows STANDARD KEYMAP IS ACTIVE or ALTERNATE KEYMAP IS ACTIVE.

To switch between keymaps during a call:

While on a call, press the Special button and enter the Switch Keymap feature code (399). The display momentarily shows STANDARD KEYMAP IS ACTIVE or ALTERNATE KEYMAP IS ACTIVE.
Automatic Call Access

This feature allows an Inter-Tel endpoint user to determine the way incoming calls are answered by entering a feature code to select the type of call access. The four variations are as follows:

- **No Automatic Call Access**: The user hears intercom dial tone when the handset is lifted and must press a Call or individual trunk button (or the \texttt{ANSWER} button) to answer an outside call, or press the \texttt{IC} button (or Call button, if there is no \texttt{IC} button) to answer an intercom call.

- **Automatic Intercom Call Access Only**: The user can answer ringing intercom calls by simply lifting the handset, but outside calls must be answered by pressing a Call button, an individual trunk button, or the \texttt{IC} button.

- **Automatic Outside Call Access Only**: The user can answer ringing outside calls by lifting the handset, but ringing intercom calls must be answered by pressing the \texttt{IC} button (or Call button, if there is no \texttt{IC} button).

- **Automatic Access For All Calls**: The user can answer any ringing call by lifting the handset. If no call is ringing, the user hears intercom dial tone when the handset is lifted. (This is the way single line endpoints work. When the system is in the default state, all Inter-Tel endpoints have this type of call access.) In all variations, users can receive handsfree intercom calls (if enabled).

The selected option determines how all types of ringing intercom or outside calls (direct calls, transferred calls, recalls, etc.) are answered. If more than one call is ringing at the endpoint, the first call received is the first answered.

When programmed for automatic outside call access, a user with allowed answer, but without ring in, for a ringing trunk must always press an individual trunk button for that trunk or enter the Automatic Trunk Answer feature code (350) to answer the incoming call. Transferred calls and recalls can be answered by lifting the handset.

Camped-on calls cannot be answered by simply lifting the handset or pressing the Speaker button. For example, an endpoint is programmed to automatically answer ringing outside calls, but requires pressing the \texttt{IC} button to answer ringing intercom calls. If a private intercom call rings in and is immediately followed by an outside call ringing in, the display shows the intercom call message, and the outside call camps on. The intercom call also camps on when the handset is lifted. The user can then choose between the camped on calls by pressing either the \texttt{IC} button or the Call or individual trunk button (or \texttt{ANSWER} button).

The automatic call access options outlined below can be programmed at Inter-Tel endpoints only. Single line endpoints are designed to automatically answer ringing intercom and outside calls by lifting the handset and cannot be changed.

**To program outside call access:**

To change the current outside call access option to its opposite (toggle on or off), enter the Automatic Trunk Access On/Off feature code (360). You hear a confirmation tone. The display shows AUTO TRNK ACCESS ON (or OFF).

**To program intercom call access:**

To change the current intercom access option to its opposite (toggle on or off), enter the Automatic Intercom Access On/Off feature code (361). You hear a confirmation tone. The display shows AUTO IC ACCESS ON (or OFF).
Music-On-Hold and Background Music

The Music-On-Hold feature not only makes waiting on hold as pleasant as possible, but it assures the holding party that the call is still connected.

Endpoint users can listen to background music by entering the Background Music feature code (313).

The system can be equipped with one optional external music sources for the Music-On-Hold feature. The music source can be a customer-provided radio, tape player, or other device connected to the Music On Hold (MOH) jack on the back of the Inter-Tel 5000 Base Server. When a music source is installed, music-on-hold is enabled or disabled in DB Programming. If music is not wanted, the system can be programmed for tick-tones or silence on hold. (The Background Music feature uses the music source connected to the unit only).

In some circumstances, there may be broadcast restrictions associated with the music heard by callers on hold. Check with the music's original distributor and/or the radio station for copyright and broadcast restrictions concerning music-on-hold.

In a network, each node has its own music source(s). If a caller on Node 1 is holding for a user on Node 2, the caller hears the music on node 2. Note that if a caller is holding for a user on Node 1 and the call moves to Node 2 (due to a transfer, forward, or recall), the user will stop hearing the music on Node 1 and will begin hearing the music on Node 2.

The music-on-hold option that will be heard by outside callers on the trunks in the trunk group can be set for music, silence, tick tones, ringback, or silence. Throughout the manual, the term “music” refers to the selected option.

Background music is interrupted for calls, pages, endpoint programming, and ringing. If you receive complaints from IP endpoint users about losing background music, there may be a firewall issue. Contact your local Inter-Tel dealer for assistance.

To turn endpoint background music on or off:

While on- or off-hook, enter the Background Music On/Off feature code (313) or press the analog endpoint’s MUSIC button. If off-hook, hang up. The display shows BACKGROUND MUSIC ON (or OFF).
Multilingual Capability

This feature requires 0 feature units, but it is installer-programmable to prevent unintentional language changes by users in an all-English system.

The system provides a choice between American English, British English, Spanish, and Japanese prompts and displays. The system selects the language to use for each call, as determined by the trunk, endpoint, and voice processing system programming, as described below. Japanese prompts can be viewed only on the digital display endpoints. Other Inter-Tel endpoints do not support Japanese displays.

Endpoints

The system can be programmed to use a Primary Language and a Secondary Language, as described in the following sections.

Language Selection

Available languages are American English, British English, Mexican Spanish, and Japanese. An endpoint flag in DB Programming determines the language that is used by each endpoint.

For example, if the endpoint is programmed for American English, all Inter-Tel endpoint displays appear in American English. Also, when the endpoint is used for calling a voice processing system application, the voice prompts are in American English (unless overridden by a Call Routing Announcement as described on page 161). If programmed for Japanese, the displays are in Japanese (Katakana) characters and the voice prompts are in Japanese (unless overridden). By default, all endpoints are set for the system’s Primary Language.

Only digital display endpoints display Japanese prompts. Other Inter-Tel endpoints do not support Japanese displays.

All displays and default messages in the system are provided in American English, British English, Japanese, and Spanish. Endpoint-programmed messages, including those programmed through the administrator’s endpoint, can include English/Spanish or Japanese characters, or a combination. In displayed lists, such as directories, the English/Spanish characters are alphabetized before the Japanese characters. Therefore, Japanese names appear at the end of the list.

When programming a feature that requires a time and date, an English/Spanish endpoint user enters the date in the order “month, day, year” and the time in the order “time, AM/PM.” The Japanese endpoint user enters the date as “year, month, day” and the time as “AM/PM, time.” The displays show the time and date differently, as well. For example, the English/Spanish display shows the time and date as 12:25 TUE NOV 28 and the Japanese display shows TUE 11/28 12:25.
Secondary Language Selection

A field titled “Secondary Language” is provided for each endpoint in the system. This field corresponds to the Change Language feature (301). This feature toggles between the System Primary Language and the endpoint Secondary Language, or it can specify a language. This allows any endpoint in the system to have its own secondary language or use the System Secondary Language, giving the system the ability to support more than two languages.

If an endpoint’s Secondary Language field is programmed as Use Primary Language, the Change Language feature will do nothing because the endpoint toggles between the System Primary Language and the endpoint Secondary Language (the System Primary Language).

If an endpoint’s Secondary Language field is programmed as Use Secondary Language, the Change Language feature will toggle between the System Primary Language and the endpoint’s Secondary Language, which is the System Secondary Language. This state is the system default.

If an endpoint’s Secondary Language field is programmed as Japanese, the Change Language feature will toggle between the System Primary Language and the endpoint Secondary Language, which is Japanese.

The Language field for endpoints indicates the language the endpoint is currently set to. This field used to toggle between the Use Primary Language and Use Secondary Language. It can be set to any specific language along with the Use Primary Language and Use Secondary Language. A side effect of changing the Language field is that if you change it to Japanese at the same time the endpoint Secondary Language field is set to Spanish, then the user will have no way to get back to Japanese if they enter the Change Language feature code. This situation occurs because the first time the user enters the feature code, the system will toggle the endpoint to the Use Primary Language. The next time the user enters the feature code, the system will toggle the language to the endpoint’s Secondary Language, which is Spanish. To avoid this situation, simply change the endpoint’s Secondary Language to Japanese and the endpoint’s Language field to Japanese.

| IMPORTANT | Setting an endpoint’s Secondary Language field to a specific language will not work across nodes unless all Axxess nodes in the system are at least 5.2F6 or later. This is because the older systems do not understand specific languages and expect to receive only Primary Language or Secondary Language values. |

Trunks

A flag in DB Programming determines the language that will be used by each trunk. The language choices are Use Primary Language, Use Secondary Language, American English, British English, Japanese, or Spanish. This field can be set to a specific language so that the system can support more than two languages. If the trunk is programmed for the Primary Language, callers who reach a voice processing system application will hear the voice prompts in the Primary Language (unless overridden by a Call Routing Announcement as described on “Voice Processing System” on page 161). If programmed for the Secondary Language, the voice prompts will be in the Secondary Language (unless overridden). By default all trunks are set for the Primary Language. See the Important note above.
Do-Not-Disturb and Reminder Messages

The system has default Do-Not-Disturb and Reminder Messages in both languages. The messages that use Japanese characters can be reprogrammed only through an administrator endpoint or DB Programming. See “System Programming” on page 36.

The language of the messages seen by the users, both when programming their endpoints and when calling another endpoint that is in Do-Not-Disturb, are determined by the endpoint’s programmed language. That is, if an endpoint is programmed for Japanese, the user will see only Japanese messages when programming a message. If a Japanese-programmed endpoint calls an English-programmed endpoint that has selected Do-Not-Disturb message 02, the Japanese endpoint user will see the Japanese version of message 02. Of the IP endpoints, only the Model 8660 can use Japanese prompts.

Only digital display endpoints will display Japanese prompts. Other Inter-Tel endpoints do not support Japanese prompts.

When Do-Not-Disturb or Reminder Messages are reprogrammed, the programmer should attempt to keep the meanings for the messages in both lists the same. That is, if the Primary Language Do-Not-Disturb message 02 is changed to “PAGE ME,” a similar message should be programmed for the Secondary Language Do-Not-Disturb message 02.

Voice Processing System

When a voice processing system application receives a call from an endpoint or trunk, the system tells the application which language is programmed for that device. For example:

- If a trunk programmed for American English rings in to an application, the voice processing system will play the American English prompts.
- If an endpoint programmed for Japanese calls Voice Mail, the user will hear Japanese prompts.
- If an endpoint programmed for American English receives a call on a trunk that is programmed for Japanese, and then transfers the call to Voice Mail, the caller (using the trunk) will hear Japanese voice prompts.
- If a Japanese endpoint is forwarded to Voice Mail, a caller on an American English endpoint or trunk will hear the “American English” prompts (the user-recorded mailbox greeting will be heard in the language in which it was recorded) when the call is forwarded.

When using a Call Routing Announcement application with digit translation, the individual voice processing system applications assigned to the digits can be programmed to override the device language and provide prompts in one language only. Using this method, you can program a Call Routing Announcement tree that offers callers a choice between languages. For example:

- The Call Routing Announcement application could have a greeting that says, “Thank you for calling. For English prompts, press 1. NIHONGO WA, 2 WO OSHITE KUDA-SAI (for Japanese, press 2).”
- The digit translation for digit 1 would lead to an application that overrides the calling device’s programming and uses only American English prompts.
- The digit translation for digit 2 would lead to an application that overrides the calling device’s programming and uses only Japanese prompts.
In the previous example, the digit translations could be nodes that lead to various other American English-only or Japanese-only applications. Or, the first level can give more choices, such as English Voice Mail or Automated Attendant and Japanese Voice Mail and Automated Attendant. Either way, the individual applications or nodes can be programmed to play only one language or they can use the device’s language, as needed.

**Multi-Language Feature on Network Operation**

Nodes do not have to use the same Primary and Secondary languages because the Voice Mail system installs all existing language prompts, not just prompts for the languages designated as Primary or Secondary. Therefore, for calls across network nodes, if the language does not match between the two nodes, the called Voice Mail system searches for the designated language in its system in the following order: Primary language, Secondary language, American English, British English, Japanese, and Spanish and uses the designated language. If the designated language does not exist in the Voice Mail system, the system uses the default Primary language instead.

**Directories**

If Multilingual Support is enabled, directory users will be able to enter English/Spanish and Japanese (Katakana) characters, as needed. Procedures for using the directories are given on page 225.

In directories, the English characters are alphabetized before the Japanese characters. Therefore, Japanese names will be at the end of the list, after the English names.

**Change Language Feature Code**

A user can change the assigned language for the endpoint by entering the Change Language feature code only when the endpoint is idle.

**To change the endpoint language:**

- **Single line endpoints:** Lift the handset and enter the Change Language feature code (301) to change your endpoint between the Primary and Secondary Language, as appropriate. To verify the current language, you can call a voice processing system application that does not override the device language.

Changing an endpoint’s language automatically changes any display. For example, if an endpoint changes from English to Spanish while in Do-Not-Disturb mode with message 02, the display would change from American English message 02 to Spanish message 02.

Only digital display endpoints display Japanese prompts. Other Inter-Tel endpoints do not support Japanese prompts.

**Intercom Calls**

When the procedures in this section tell you to hookflash [recall], quickly press and release the hookswitch. If you press the hookswitch to hang up, hold it down for a couple of seconds. Otherwise, the system recognizes it as a hookflash [recall].

The intercom (IC) feature allows the user to place endpoint-to-endpoint calls that can be answered either in handsfree mode or in private, or non-handsfree, mode. A user that reaches a busy endpoint can camp on, request callback queue, or leave a message. As described in this section, other features that apply to both outside and IC calls include call waiting, call transfer, reverse transfer, call forwarding, and placing calls on hold.

The network allows handsfree intercom calls when calling from one node to another, unless the call camps on to the node before being sent to the other node.
Programming for Private Intercom Calls

A user places private IC calls by pressing the pound (\#) button before dialing the extension number. Or, the endpoint can be programmed to always send and/or receive private calls using the Ring Intercom Always or Handsfree Disable features.

Ring Intercom Always

An endpoint user can always place private calls by programming the endpoint with the Ring Intercom Always feature code. While this feature is enabled, the called party hears repeating double tones. To answer, the called party must lift the handset or press the Speaker button, ANSWER, or IC button. If the endpoint does not have an IC button, the call will appear under a Call button. In the default database, all single line endpoints have this feature enabled.

To turn ring intercom always on or off:

Single line endpoints: Lift the handset. Enter the Ring Intercom Always On/Off feature code (377).

Handsfree Enable/Disable

An Inter-Tel endpoint user can always receive private calls by disabling the endpoint Handsfree Answering feature using the Handsfree On/Off feature code, as described in the following paragraph. The endpoint user hears repeating double tones when receiving an intercom call. To answer while Handsfree Mode is disabled, the user must lift the handset or press the Speaker, ANSWER, or IC button.

To enable or disable handsfree mode:

While on-hook, enter the Handsfree On/Off feature code (319). The display shows HANDSFREE MODE ON (or OFF).

Endpoint-To-Endpoint Calling

The following sections describe how to receive and place intercom calls on Inter-Tel multi-line endpoints and on single line endpoints.

Receiving Intercom Calls

To receive an intercom call on an Inter-Tel endpoint:

If you hear a double tone and a call announcement (handsfree call), and the Speaker button is lit, respond handsfree or lift the handset. (Display endpoints show IC TO <USERNAME> to identify the source of the call.)

If you hear repeating double tones (private call), lift the handset, or press the Speaker, ANSWER, or IC button to respond handsfree. (Display endpoints show INTERCOM CALL FROM <USERNAME> to identify the source of the call.)

Model 8660 or 8662: Instead of answering the call, you may press one of the following menu buttons:

* SEND TO V-MAIL: This option appears only if you have a Voice Mailbox. Pressing this button sends the call directly to your mailbox greeting.

* IC CALL TO DND: Pressing this button sends Do-Not-Disturb tones to the caller and allows you to program a Do-Not-Disturb message. If using a display endpoint, the caller sees DO-NOT-DISTURB. For detailed instructions about using Do-Not-Disturb, see page 238.
To receive an intercom call on a single line endpoint:
When you hear repeating double tones, lift the handset.

To receive a transferred-to-hold intercom call:

**Inter-Tel endpoints:** If the endpoint does not have an \[IC\] button, the call will appear under a Call button. When the \[IC\] button (or Call button) is fluttering and the display shows \(<\text{calling endpoint}>\) TFR FROM \(<\text{transferring endpoint}>\), lift the handset and press the \[IC\] button (or Call button). Or, while on-hook, press the \[IC\] button (or Call button) to speak handsfree. You are connected to the caller. Executive Display, Professional Display, and Model 8560 endpoint users have the same menu button options as described for a ringing intercom call.

**Single line endpoints:** When you hear a page announcing that a call is holding at your endpoint, lift the handset (at your endpoint) and enter the Individual Hold feature code (336). You are connected to the caller.

### Placing Intercom Calls

When placing an intercom call, begin dialing before the Dial Initiation timer expires. If the timer expires, the system drops the connection and sends reorder tones.

**To place an intercom call:**

1. For all Inter-Tel endpoints:
   - **To use the handset:** Lift the handset. If the endpoint has a display it shows ENTER EXTENSION NUMBER. If you want to place a private (non-handsfree) call to an Inter-Tel endpoint, press the pound (\#) button before dialing the extension number.
   - **To use the speakerphone:** Skip to step 2 (begin dialing). The Speaker button lights. The display shows ENTER EXTENSION NUMBER.

**Executive Display, Professional Display, and Model 8660 or 8662:** Lift the handset or begin dialing as described above. The display shows ENTER EXTENSION NUMBER. You may press one of the following menu buttons:

- IC DIRECTORY: Press this button to use the Intercom Directory feature. See page 225 for instructions.
- OUTSIDE CALL: If you would rather make an outside call, press this button and see the instructions on page 178.

**Single line endpoints:** Lift the handset. (If the DTMF decoders are busy, you hear silence. Hang up and try again.)
2. Dial an extension number. The number can be 0 for the attendant, an endpoint extension number, or a hunt group pilot number. Display endpoints show the number dialed. If you dial too slowly, the Inter-digit timer may expire and you will hear reorder tones. One of the following will occur:

   a. **If calling a handsfree Inter-Tel endpoint**, speak after you hear a double tone. The display shows IC TO &lt;username&gt;.

   b. **If placing a private call or calling a single line endpoint, a hunt group, or an Inter-Tel endpoint with handsfree disabled**, you hear repeating double tones until the call is answered. The display shows CALL RINGING FOR &lt;username&gt;. When the call is answered, the display shows IC TO &lt;username&gt;. Executive Display, Professional Display, and Model 8560 endpoint users can press one of the following menu buttons while the call is ringing:

      - LEAVE MESSAGE: This button appears only if you have called a station. Press this button to leave a message waiting indication at the station. See page 167 for details.

      - LEAVE VOICE MAIL: This button appears only if the called station has a Voice Mailbox. Press this button to call the station’s Voice Mailbox.

   c. **If you dial an invalid number**, you hear reorder tones. Display endpoints show INVALID NUMBER. If you do not have an assigned attendant and you dialed 0, the display shows NO ATTENDANT PROGRAMMED. If you dialed a hunt group number with no assigned endpoints, the display shows NO MEMBERS IN HUNT GROUP.

   d. **If the called endpoint is busy**, the display shows &lt;username&gt; IS BUSY. You have the following options (in addition to the menu buttons described previously):

      - **Disconnect**: Hang up and try later. If dialing handsfree (Inter-Tel endpoints only), press the Speaker button to disconnect.

      - **Camp On**: Do not hang up, your call camps on. When the called endpoint is available, a private call is placed. See page 166 for details.

      - **Request a callback (queue)**: This is similar to camping on except that you hang up and wait for the busy endpoint to become available. (For more information, see page 166.) Executive Display, Professional Display, and Model 8560 endpoint users can press the QUEUE menu button.

   e. **If you hear a repeating signal of four fast tones and a pause**, the endpoint is in Do-Not-Disturb. You can use the Queue Callback feature or, if enabled on your endpoint, use the Do-Not-Disturb Override feature to place a nonhandsfree call to the endpoint by pressing the Special button and entering the Do-Not-Disturb Override feature code (373) or pressing the OVERRIDE DND menu button.

   f. **If the called endpoint is busy, in Do-Not-Disturb, or there is no answer**, you can leave a message waiting indication with the endpoint itself or with the endpoint’s message center. For details, see page 167. Executive Display, Professional Display, and Model 8560 endpoint users can press the LEAVE MESSAGE or LEAVE VOICE MAIL menu button.

   g. **If the called station is a busy endpoint, and if all the necessary conditions for OHVA are enabled**, you can complete the call using the Off-Hook Voice Announce feature. For a complete explanation of this feature, see “Off-Hook Voice Announce (OHVA)” on page 176.
Intercom Camp On and Queue Callback

Camp On and Queue Callback requests allow the caller to wait for an available resource. A user waiting for a specific resource (a trunk or an endpoint) will be served before a user waiting for a group that contains that specific resource (trunk group or hunt group). This includes camped on Emergency Call feature calls—if the Emergency Call feature uses a trunk group, requests for specific trunks will be processed ahead of the emergency call.

Camp On

When a user calls a busy endpoint or hunt group, the system sends a busy signal. The caller can wait off-hook to camp on (after the Camp On timer expires) and hear music while waiting until the called endpoint is available. The system periodically sends Call Waiting signals to the busy endpoint(s).

A user can camp on to busy endpoints on other nodes and will hear the other node’s music-on-hold while camped on.

If an endpoint enables Do-Not-Disturb while an intercom call is camped on, the camped-on caller is removed from the camped on state and that caller receives Do-Not-Disturb indications (see page 189). Intercom callers cannot camp on to an endpoint that is in Do-Not-Disturb.

To camp on to an extension number:

- **All Inter-Tel endpoints**: If you hear a busy signal when calling an extension number and want to camp on, do not hang up. If Music-On-Hold is enabled, you hear music while camped on. When the endpoint is available, a private call is automatically placed, and you hear repeating double tones until the call is answered. The display shows WAITING FOR <username>.
- **Model 8660 or 8662**: You may press one of the following menu buttons:
  - LEAVE MESSAGE: This button appears only if you have called an endpoint. Press this button to leave a Message Waiting indication at the endpoint. See page 167 for details.
  - LEAVE VOICE MAIL: This button appears only if the called endpoint has a Voice Mailbox. Press this button to call the endpoint’s Voice Mailbox.
  - QUEUE: Press this button to request a callback when the resource is available, as described in the following section.

Busy Endpoint Callback Queue

When a called endpoint is busy or in Do-Not-Disturb, the caller can apply the Queue Callback feature and hang up until the endpoint becomes available. This can be done even if the call camps on. Each endpoint can apply only one active Queue Callback request at a time.

The Queue Callback feature can also be used when calling a hunt group if all endpoints are unavailable. A user cannot queue onto busy resources on another node.

Queue callbacks must be answered before the Queue Callback timer expires. If a callback is not answered, the queue is canceled. If the endpoint is busy when called back, the queue request is placed behind any other waiting queue requests. Queue callbacks do not follow forwarding and are not blocked by Do-Not-Disturb mode.
To request a callback from an endpoint:

1. **Inter-Tel endpoints**: When you hear a busy signal or Do-Not-Disturb signal (a repeating signal of four fast tones and a pause) when calling an extension number, press the QUEUE menu button. Or press the Special button, enter the Queue Callback feature code (6), and hang up. The display shows QUEUE REGISTERED FOR <username>.

   **Single line endpoints**: When you hear a busy signal or Do-Not-Disturb signal when calling an extension number, hookflash and enter the Queue Callback feature code (6). Then hang up.

2. When the queued endpoint is available, your endpoint will ring. Display endpoints show <username> IS NOW AVAILABLE. Lift the handset or press the ANSWER button. A private call is automatically placed to the queued endpoint.

   **Executive Display, Professional Display, and Model 8660 or 8662**: You may press one of the following menu buttons:
   - CANCEL QUEUE: To cancel the queue request, press this button without lifting the handset.
   - ANSWER: You may press this button instead of lifting the handset to answer the queue callback using the speakerphone.

To cancel a queue request before the queued endpoint becomes available:

- **Inter-Tel endpoints**: While on-hook, enter the Queue Callback feature code (6). The display shows QUEUE REQUEST CANCELED.
- **Single line endpoints**: Lift the handset, enter the Queue Callback feature code (6), and hang up.

### Call Logging

The Call Logging feature lists the following types of calls for users of Inter-Tel digital and IP endpoints and Inter-Tel IP softphone applications:

- Missed Calls
- Received Calls
- Dialed Calls

The Call Logging feature helps you see who called when you were away and makes it easy to redial those people. Non-display and single line endpoints do not support this feature. Six-line display endpoints are recommended for field visibility and ease of use.

Benefits of Call Logging include the ability to redial intercom (IC) and outgoing calls, to store Caller ID data from incoming calls, and to identify missed calls. The system provides you with an interface similar to cell phones. Call lists are stored in Call Processing and are accessible through the user interface screen on display endpoints.
Call Logging Display

The following graphic shows an example of the Call Logging display on a six-line display endpoint. Two-line display endpoints show only the top two lines. Each call entry contains the following fields:

- **Name**: Displays the name of the missed/received/dialed call. If not available, UNAVAILABLE is displayed.
- **Number**: Displays the number of the missed/received/dialed call. If not available, UNAVAILABLE is displayed.
- **Date/Time**: Displays the date and time that the call was missed/received/dialed.
- **DELETE/DELETE ALL**: Allows the user to delete the entry indicated on the display or to delete all entries.
- **CALL NOW**: Allows the user to call the number indicated on the display.
- **ARROW MENU BUTTONS (<< and >>)**: Appear on a six-line display endpoint if more than one entry exists on a Call Log list.

A maximum of 20 entries can be stored in each of the three Call Logs associated with an endpoint. Call Log entries are displayed from newest to oldest. After a Call Log reaches its maximum number of entries, the next call entry appears at the top of the list and the oldest call entry is deleted. Using arrow menu buttons on the six-line display endpoints or Volume Up and Volume Down buttons, you can scroll through all the entries in a Call Log list. For details, see “Call Logging Displays on a Six-Line Display Endpoint” on page 169 and “Call Logging Displays on a Two-Line Display Endpoint” on page 171.
Call Logging Displays on a Six-Line Display Endpoint

**NOTE** At any menu level, you can press the asterisk (*) button to cancel or return to the previous menu, or press the pound (#) button to accept.

The following examples show the Call Logging displays that appear on a six-line endpoint, which is recommended for system administrators.

**Idle Menu Display**

An option on the Idle Menu of six-line endpoints allows you to access the logs directly. Note that the six-line administrator display endpoints no longer show **ADMIN FEATURES** on the first line. Rather, it shows only **ADMIN**, which makes room for the **MISSED XX** (where XX can be 1–99) calls menu option on the right side of the screen.

<table>
<thead>
<tr>
<th>Non-Admin Endpoint</th>
<th>Admin Endpoint</th>
</tr>
</thead>
<tbody>
<tr>
<td>29999 JOHN DOE</td>
<td>29999 JOHN DOE</td>
</tr>
<tr>
<td>1:53 FRI AUG 20</td>
<td>1:53 FRI AUG 20</td>
</tr>
<tr>
<td>OUTGOING DND</td>
<td>OUTGOING DND</td>
</tr>
<tr>
<td>DIRECTORY LOGS</td>
<td>DIRECTORY LOGS</td>
</tr>
<tr>
<td>OTHER FEATURES</td>
<td>OTHER FEATURES</td>
</tr>
</tbody>
</table>

<table>
<thead>
<tr>
<th>Non-Admin Endpoint</th>
<th>Admin Endpoint</th>
</tr>
</thead>
<tbody>
<tr>
<td>JOHN DOE 4809619000</td>
<td>JANE DOE x29999</td>
</tr>
<tr>
<td>FR AUG 20 11:53A</td>
<td>FR AUG 20 11:53A</td>
</tr>
<tr>
<td>DELETE DEL ALL</td>
<td>DELETE DEL ALL</td>
</tr>
<tr>
<td>CALL NOW &lt;&lt;</td>
<td>CALL NOW &lt;&lt;</td>
</tr>
</tbody>
</table>

**Main Menu**

When the LOGS menu option is selected, the following display appears.

- **LOG TYPE**: MISS=1
- RCV=2 DL=3 CLR=4
- MISSED CALLS
- RECEIVED CALLS
- DIALED CALLS
- CLEAR CALL LOGS

**Call Log Display**

When one of the call menu options (Missed=1, Received=2, or Dialed=3) is selected in the main menu, a display similar to the one below appears. The arrow (<< and >>) menu buttons at the bottom of the display appear only if more than one entry appears in the list on a six-line display endpoint.

<table>
<thead>
<tr>
<th>Non-Admin Endpoint</th>
<th>Admin Endpoint</th>
</tr>
</thead>
<tbody>
<tr>
<td>JOHN DOE 4809619000</td>
<td>JANE DOE x29999</td>
</tr>
<tr>
<td>FR AUG 20 11:53A</td>
<td>FR AUG 20 11:53A</td>
</tr>
<tr>
<td>DELETE DEL ALL</td>
<td>DELETE DEL ALL</td>
</tr>
<tr>
<td>CALL NOW &lt;&lt;</td>
<td>CALL NOW &lt;&lt;</td>
</tr>
</tbody>
</table>

OR

<table>
<thead>
<tr>
<th>Non-Admin Endpoint</th>
<th>Admin Endpoint</th>
</tr>
</thead>
<tbody>
<tr>
<td>JOHN DOE 4809619000</td>
<td>JANE DOE x29999</td>
</tr>
<tr>
<td>FR AUG 20 11:53A</td>
<td>FR AUG 20 11:53A</td>
</tr>
<tr>
<td>DELETE DEL ALL</td>
<td>DELETE DEL ALL</td>
</tr>
<tr>
<td>CALL NOW &lt;&lt;</td>
<td>CALL NOW &lt;&lt;</td>
</tr>
</tbody>
</table>
Scrolling Through Entries

To scroll through the entries, press the >> menu button or the Volume Up button to scroll forward, or press << or the Volume Down button to scroll backward. If no previous page or entry exists, the << menu button is unavailable and the Volume Down button is unresponsive. Likewise, if no next page or match exists, the >> menu button is unavailable and the Volume Up key does not respond.

Deleting Entries

To delete the entry, press DELETE or 0. The display shows the next entry in the list or returns to the main menu if the list is empty.

To delete all entries, press DEL ALL or 4.

Placing a Call

To place a call, press CALL NOW or the pound (#) button.

Missed Calls

The Missed Calls feature is supported only on six-line display endpoints. If an endpoint has registered missed calls that the user has not yet viewed, the MISSED XX (where XX is 1–99) calls menu option appears on the display screen, as shown in the following example.

When the user selects the MISSED XX calls menu option from an idle display, the user is taken directly to the MISSED Calls menu, bypassing the LOGS menu.

If using a six-line display endpoint in conjunction with the Unified Communicator (UC) application in a network environment, missed calls may be registered accurately only in the UC Call Log. Missed calls that have been routed across nodes may show as Received rather than Missed. In this situation, the user should rely on the UC Call Log for an accurate view of missed calls.

After selecting the Missed Calls menu option, the Idle Display menu no longer shows MISSED XX on the display, regardless of whether or not the user actually looks at every one of the missed calls.

To access the Missed Calls log at any time, the user can either press the LOGS menu button next to the endpoint display screen or press the Special button and enter the Call Logging feature code (333). The screen displays the LOGS menu, and from there the user can navigate to the Missed Calls menu.
Call Logging Displays on a Two-Line Display Endpoint

At any menu level, you can press the asterisk (*) button to cancel or return to the previous menu, or press the pound (#) button to accept.

The following examples show Call Logging displays that appear on a two-line endpoint.

Main Menu

If you have a two-line display endpoint, use a feature code to access the Call Logs. When you press the Special button (SPCL or infinity symbol) then enter the Call Logging feature code (333), the following display appears. It prompts you to specify which Call Log you want to access or if you want to clear all call logs.

```
LOG TYPE: MISS=1
RCV=2 DL=3 CLR=4
```

Call Log Display

When one of the call menu options (Missed=1, Received=2, or Dialed=3) is selected in the main menu, a display similar to the one below appears.

```
4809619000 JOHN
FR AUG 20 11:53A
```
If both the calling number and Caller ID name are available, the first line contains the number and as much of the name as possible. If neither the calling number nor Caller ID name is available, UNKNOWN CALLER is displayed.

```
4809619000
FR AUG 20 11:53A
```
If the calling number is available, but the Caller ID name is not available, the first line contains only the calling number.

Scrolling Through Entries

To scroll through the entries, press the Volume Up button to scroll forward, or press the Volume Down button to scroll backward.

Deleting Entries

To delete the entry, press 0. The display shows the next entry in the list or returns to the main menu, if the list is empty.

To delete all entries, press 4

Placing a Call

To place a call, press the pound (#) button.
Station-to-Station Messages

Intercom callers may leave a message waiting indication if a called endpoint is busy, if there is no answer, if the endpoint is in Do-Not-Disturb, or when they are connected to or placed on hold by another endpoint. Users are allowed to leave message for, and respond to messages from, users on other nodes. There are two messaging options:

- **Have the called party return your call:** Display endpoints show the source and time of the message. When the called party responds to the message indication, a call is automatically placed to your endpoint.

- **Leave a message with the called party's message center:** Display endpoints show that a message was left with the endpoint user’s assigned message center. When the called party responds to the message indication, a call is automatically placed to his or her message center.

To signal that a message is waiting, the called Inter-Tel endpoint button flashes, and the display shows the number of waiting messages. Each time the button is pressed, the display shows the message source for each of the waiting messages in the order they were received. For messages from endpoints, the display shows MSG: <endpoint> and the date and time of the message. For messages from Voice Mailboxes, the display shows XX MESSAGES FROM MBOX XXXX to indicate the number of waiting messages in each mailbox that left a message. (This is especially helpful at endpoints with multiple mailboxes.) For single line endpoints, a system programming option can be enabled that sends six short message waiting tones when the user lifts the handset or presses the hookswitch.

IP and digital endpoints are equipped with a Message Lamp located on the upper-right corner of the endpoint. If programmed for the Message Waiting feature, this lamp flashes/lights whenever the button flashes/lights. For details, see page 113.

Any endpoint, voice processing system application, hunt group, or off-node device can be designated as the message center for an endpoint. However, an endpoint cannot be programmed as its own message center.

If the designated message center is a Voice Mail hunt group, the Voice Mail hunt group is called after the Message Wait timer expires. When the Voice Mail unit answers the call, the called party’s “mailbox” is automatically dialed. The caller can then leave a message in the mailbox. The called party’s button flashes, and the message display indicates that the message was left by the Voice Mail hunt group.

Each endpoint user can leave message waiting indications at more than one endpoint. If more than one message is received from one endpoint, the message display shows only the first message, and all other requests are ignored.

When an endpoint is forwarded or has system forwarding, and a caller leaves a Message Waiting indication after calling the forwarded endpoint, the message indication appears at the original endpoint instead of the one that received the forwarded call.
Leaving Messages

To leave a message waiting indication:

1. When calling an endpoint, if you hear a busy signal, Do-Not-Disturb signal (repeating signal of four fast tones and a pause), or the call is not answered:

   **Inter-Tel endpoints**: Press the MSG button. Or press the Special button and enter the Message feature code (365). The endpoint receives a Message Waiting indication. The display shows HANG UP OR WAIT FOR MESSAGE CENTER.

   **Single line endpoints**: Hookflash [recall] and enter the Message feature code (365).

2. To have the called party call you: Hang up or press a trunk button or the IC button to disconnect before the Message Wait timer expires. The display shows MESSAGE LEFT FOR <username>.

To leave the message with the called party's message center: Do not hang up. When the Message Wait timer expires, a private call is automatically placed to the called party’s message center. (If the message center is a Voice Mail hunt group, the called party’s mailbox is automatically dialed after the Voice Mail unit answers the call.) Leave your message with the message center and hang up. The called endpoint receives message waiting indications. If the message center does not answer the call (that is placed after the timer expires), the display will show that the message was from your endpoint and you will be called when the endpoint user responds to the message indications. The display shows MESSAGE LEFT FOR <username>.

If the endpoint you are calling does not have a designated message center, you will hear dial tone immediately after pressing the MSG button. The endpoint will receive a Message Waiting indication.

Silent Messages

Silent messages can be placed while on- or off-hook without making an intercom call to the endpoint. This method should be used by analog Voice Mail units.

To leave a silent message:

**Executive Display, Professional Display, and Model 8660 or 8662 endpoints**: While on-hook, press the MSG button. Then press the LEAVE MESSAGE menu button. The display shows LEAVE MESSAGE ON EXTENSION. Enter the extension number you want. The number is shown on the display. (If necessary, you can press the IC DIRECTORY menu button to use the intercom directory to look up a number. For instructions, see page 225.)

**Other Inter-Tel endpoints**: While on-hook, enter the Silent Message feature code (367). Then dial the extension number you want.

**Single line endpoints**: Lift the handset and enter the Silent Message feature code (367). Then dial the extension number you want and then hang up.
Canceling a Message Waiting Indication

Occasionally, before the person you called responds to your message, you may want to cancel a Message Waiting indication that you left on their endpoint. If the message was left with the person’s message center, you cannot cancel the Message Waiting indication; only the message center can cancel that message. Use the following procedure to cancel a message that you left on an endpoint.

To cancel a message you left:

1. Executive Display, Professional Display, and Model 8660 or 8662: While on-hook, press the \[MSG\] button. Then press the \[CANCEL MESSAGE\] menu button. The display shows \[CANCEL MESSAGE ON EXT\#. If you want, you can press the \[IC DIRECTORY\] menu button to use the intercom directory to look up a number. See page 225 for instructions.

   Other Inter-Tel endpoints: While on- or off-hook at your endpoint, enter the Cancel Message feature code (366). The display shows \[CANCEL MESSAGE ON EXT \#\].

   Single line endpoints: Lift the handset and enter the Cancel Message feature code (366).

2. Enter the extension number of the endpoint that has the message indication that you want to cancel. You hear intercom dial tone if off-hook, or confirmation tone if on-hook. The display shows \[MESSAGE CANCELED FOR <username>\]. If a message was not present at the called endpoint, you hear reorder tones and the display shows \[NO MESSAGES TO CANCEL\].

Responding to Messages

Once a message has been received at an endpoint from the local voice mail unit or alternate message source, a user views the message, answers the message, and then cancels the Message Waiting indicator. In the following sections the role of the alternate message source is described along with procedures for viewing, answering, and canceling a waiting message.

Alternate Message Source

The Alternate Message Source feature enables a voice mail unit to leave messages through an endpoint location or hunt group that is designated as its alternate message source. When a message is left on the voice mail unit for a user who has an assigned mailbox, the called endpoint receives the Message Waiting indications. If it is a display endpoint, the display shows \[MESSAGE RECEIVED FROM <alternate message source>\] rather than from the voice mail circuit. When the user retrieves the message, an intercom call is placed to the alternate message source.

An endpoint’s message center or alternate message source does not need to be on the same node as the endpoint. It can be an endpoint, hunt group, Voice Processing application, or any off-node device.
Viewing Waiting Messages

To view received messages at a display endpoint:

1. If your MSG button is flashing, one or more messages are waiting and the display shows YOU HAVE XX MESSAGES.
   
   **Model 8660 or 8662**: While on-hook, press the VIEW MESSAGES menu button. Then press PREVIOUS or NEXT to scroll to the appropriate message.
   
   **Other Inter-Tel endpoints**: While on-hook, press the MSG button repeatedly to view waiting messages.

2. A displayed message will show MESSAGE RECEIVED FROM EXT XXXX. If necessary, you can answer or cancel a message while it is displayed, as described in “Answering Messages” on page 175.

Answering Messages

To answer messages at a display endpoint:

1. When the message is displayed (as described in “Leaving Messages” on page 173), press the * button or the REPLY menu button while on- or off-hook. Depending on how the message was left, an intercom call is placed to the endpoint that left the message or to your message center.

2. If the MSG button continues to flash, there is another message (display shows the next message).

   If, when you respond to the message, the called endpoint is unavailable (does not answer a private call, is busy, or is in Do-Not-Disturb), your MSG button continues to flash (and the display indicates a waiting message) unless you immediately leave a message before hanging up or cancel the message after hanging up. The message waiting indications are canceled automatically only when you have actually connected with the called endpoint. Message indications from Voice Mail will only be cleared if you save or delete the waiting Voice Mail message. The MSG button will re-light if you clear the message indication without saving or deleting the Voice Mail.

   Repeat step 1.

To answer messages at a non-display endpoint or single line endpoint:

1. **Inter-Tel endpoints**: If you see a flashing endpoint MSG button, press the MSG button. Depending on how the message was left, an intercom call is placed to either the endpoint that left the message or your message center.

   **Single line endpoints**: If you hear single line message waiting tones (six tones) after lifting the handset or hookflashing when the endpoint is idle, enter the Message feature code (365). Depending on how the message was left, an intercom call is placed to either the endpoint that left the message or your message center.

   If the called endpoint does not answer, is busy, or is in Do-Not-Disturb, you may hang up or leave a message at that endpoint before hanging up. You will continue to receive message waiting indications (MSG button flash or tones), unless you left a message while attempting to retrieve the message or canceled the message after hanging up. The message waiting indications are canceled when you are actually connected with the called endpoint. Message indications from Voice Mail will only be cleared if you save or delete the waiting Voice Mail message. The MSG button will re-light if you clear the message indication without saving or deleting the Voice Mail message.

2. If message waiting indications continue, there is another message. Repeat this procedure.
Canceling a Waiting Message

Message indications from Voice Mail will be cleared only if you save or delete the waiting Voice Mail message. The \textit{MSG} button will re-light if you clear the message indication without saving or deleting the Voice Mail message.

\textbf{To cancel a displayed message without answering at an Inter-Tel endpoint:}

1. When the message you want to cancel is displayed (as described in “Canceling a Message Waiting Indication” on page 174), remain on-hook and press the asterisk (\texttt{*}) button or the \textit{DELETE} menu button. The message is canceled.

2. If the \textit{MSG} button continues to flash, there is another message. The display shows the next message. Either respond to it or cancel it.

\textbf{To cancel a message without answering at a single line endpoint or without displaying messages at an Inter-Tel endpoint:}

1. If you receive message waiting indications, enter the Cancel Current Message feature code (368). The message is canceled.

2. If you continue to receive message waiting indications, there is another message. Either respond to it or cancel it.

\textbf{Off-Hook Voice Announce (OHVA)}

The Off-Hook Voice Announce (OHVA) feature allows an intercom caller (either single line or Inter-Tel endpoint) to establish voice contact with the called Inter-Tel endpoint user, even though the user already has a call in progress on the handset. The called party can then talk freely on both connections, using the handset for the original call and the speakerphone for the OHVA intercom call. \textit{The caller on the handset call will be able to hear the OHVA call.}

In a network, intercom callers can establish OHVA calls to Inter-Tel endpoints on other nodes.

A PC Data Port Module (PCDPM) must be installed on a digital display endpoint to provide the secondary voice path needed for the OHVA feature.

IP and SIP endpoints \textit{do not support} OHVA.

To enable the OHVA, the following must be completed:

- The OHVA system flag must be enabled. When the system is in the default state, this flag is enabled.
- Each OHVA-equipped Inter-Tel endpoint can be programmed for placing and/or receiving OHVA calls. By default, OHVA is disabled at the station level.
- Other Inter-Tel endpoints and single line endpoints can be programmed for placing OHVA calls, as necessary. Inter-Tel endpoints without PC Data Port Modules or secondary voice paths, and single line endpoints cannot receive OHVA calls since they do not have secondary voice path circuitry. By default, OHVA is disabled at the station level.
- Mini-DSS-equipped digital display endpoints can be programmed to override the Camp On and OHVA Screening timers and make Immediate OHVA calls. This is a useful option for attendant stations.

Any combination of OHVA-enabled and -disabled stations is available through DB Programming. For example, all attendants may be programmed to make, but not receive, OHVA calls, while other stations are allowed to receive, but not place, OHVA calls.
OHVA calls cannot be processed if the secondary voice path or the speakerphone of the called endpoint is not available. This occurs when the endpoint has a different OHVA call in progress, has an active data call in progress, is on an active handsfree intercom or outside call, has handsfree disabled, has a call being changed from handset to speakerphone, has a headset enabled, or is in Do-Not-Disturb. Also, OHVA calls are not possible if the caller is placing a private intercom call or has the Ring Intercom Always feature enabled.

If you press the **MUTE** button while speaking to an OHVA caller on the speakerphone, the handset microphone will be muted and the caller on the handset will not hear you or the OHVA caller. When you press **MUTE** again, the handset microphone is re-enabled.

**To place an OHVA call:**

1. When placing an intercom call to a busy OHVA-enabled Inter-Tel endpoint, do not hang up. You hear busy signals until the Camp On and OHVA Screening timers expire. If you hear music after the Camp On timer expires, conditions for an OHVA call were not met and you are camped-on. See page 166.

2. When the OHVA Screening timer expires—and if the endpoint’s secondary voice path is available—you are automatically connected to the called party’s speakerphone.

**To place an immediate off-hook voice announce call (Inter-Tel endpoints with Mini-DSS buttons only):**

This method can be used only if the database has been programmed to allow Inter-Tel endpoints with Mini-DSS buttons to bypass the Camp On and OHVA Screening timers.

1. When an intercom call is placed, using a Mini-DSS button, to a busy OHVA-enabled Inter-Tel endpoint, do not hang up.

2. Press the Mini-DSS button for that endpoint again. If the endpoint’s secondary voice path is available, you are immediately connected and may speak.

**To receive an OHVA call (OHVA-enabled Inter-Tel endpoints only):**

1. While on a call using the handset, you hear a camp-on tone. The display shows **CALL ANNOUNCE FROM <username>**. Do nothing. When the OHVA Screening timer expires, you hear a double tone. You are connected with the intercom caller via the speakerphone. Your original call remains connected on the handset.

The system provides the following language options: American English, British English, Spanish, and Japanese prompts and displays. Administrators must know which languages are considered Primary and Secondary for the system.

If you press **MUTE** while speaking to the caller on the speakerphone, the handset microphone will be muted and the caller on the handset will not hear you or the OHVA caller. Press **MUTE** again to enable the handset.
2. **To terminate the OHVA call**: Press the lit Speaker button or have the OHVA caller hang up. If you terminate the original call by hanging up the handset, you remain connected to the OHVA call in the handsfree intercom mode.

**To block an off-hook voice announce call (OHVA-enabled Inter-Tel endpoints only):**

1. While on a call using the handset, you hear a camp-on tone. The display shows **CALL ANNOUNCE FROM <username>**. To cause the intercom call to camp on, press the Speaker button. The **IC** button flashes.

2. **If you want to send Do-Not-Disturb signals to the camped-on call**, press the **DND** button.

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**Outside Calls**

When a trunk is selected for receiving or placing an outside call, the voice channel is seized and cannot be used by any other endpoint (unless the Conference feature is used, see page 195). If the trunk wanted is busy, the user can camp on or request a queue callback. Other features that apply to both outside and intercom calls are discussed later in this section of the manual. They include placing calls on hold, call waiting, call transfer, reverse transfer, conferencing, and call forwarding. See page 139 for an explanation of outgoing-access, allowed-answer, and ring-in assignments.

**Placing Outside Calls**

Inter-Tel endpoints can be programmed to select a trunk, a trunk group, or ARS when a Call button is pressed. In the default state, all Call buttons are programmed for ARS access.

On display endpoints, the dialed number is displayed with hyphens separating the toll field, equal access field, area code, office code, hookflashes, pauses, asterisks, pounds, Centrex codes, and/or absorbed digits. When the system absorbs local trunk digits, the digits are displayed even when they are not dialed. (For example, if 423 is absorbed and 6767 is dialed, 423-6767 is displayed.)

When placing a call, begin dialing before the Dial Initiation timer expires. If the timer expires, the system drops the trunk connection and sends repeating reorder tones. This prevents a trunk from being tied up accidentally.

**To place an outside call:**

1. Lift the handset. (If using an Inter-Tel endpoint, you can dial on-hook.)

2. Select an outgoing trunk, using one of the methods described below. The associated individual trunk or Call button flashes slowly. (If on-hook, the Speaker button is lit.). You must have outgoing access to a trunk to seize it.

   - **Model 8660 or 8662**: Press the **OUTGOING** menu button.
   - **Any Inter-Tel endpoint**: Select a trunk using one of the following methods:
     1. ) Press an individual trunk button for direct access.
     2. ) Enter a trunk group access code (92001–92208) or press a trunk group button. This selects a trunk in the chosen trunk group. (If the group has not been programmed, the display shows **NO TRUNKS IN TRUNK GROUP**.)
     3. ) If your endpoint is programmed to select a trunk group when a Call button is pressed, press a Call button for outgoing access.
4.) See the Automatic Route Selection (ARS) procedure page 180.

5.) Enter the Outgoing Calls feature code (8). This automatically selects an outgoing trunk according to the programmed outgoing access mode for your endpoint.

- **Single line endpoints**: Select a trunk using one of the following methods:
  1.) Enter the Outgoing Call feature code (8). This automatically selects an outgoing trunk according to the programmed outgoing access mode for your endpoint.
  2.) Enter a trunk group access code (92001–92208).
  3.) See the Automatic Route Selection (ARS) procedure on page 180.

3. You hear one of the following signals:

- **Outside dial tone**: Manually dial, speed dial, or redial the phone number you want or use the outside directory. (Display endpoints show the numbers dialed. When dialing is ended and the call is determined to be valid, the display shows the elapsed time and call cost.) If you hear repeating reorder tones, you dialed a restricted number. Also, if you dial too slowly and the Interdigit timers expire, the call may be dropped, and you will hear repeating reorder tones. If using an Inter-Tel endpoint that is on-hook, you may lift the handset to speak privately. Go to **step 4**. On Inter-Tel endpoints, an individual trunk or Call button flashes slowly and the display shows the call cost and elapsed time.

Model 8660 or 8662 endpoint users may press one of the following menu buttons:

- OUTSIDE DIR: This button appears only if you are using ARS. Press the button to use the outside directory as described on page 225.
- STN SPEED DIAL: Press this button to use an Station Speed Dial number as described on “Station Speed Dial” on page 219.
- SYS SPEED DIAL: Press this button to use a System Speed Dial number as described on page 217.
- REDIAL: Press this button to redial the last number dialed (or saved) as described on page 233.

- **Single progress tone**: Enter a forced account code using the dialpad buttons or a pre-programmed Speed Dial button (see page 184). The display shows ENTER ACCOUNT CODE and the digits as they are entered. You hear outside dial tone when the code is accepted. On Inter-Tel endpoints, an individual trunk or Call button flashes slowly.

- **Repeating reorder tones**: You are restricted from the trunk group chosen; choose another trunk group. Or, you are restricted to using ARS only; try again using ARS (see page 180). The display shows OUTGOING ACCESS DENIED.

4. After dialing the call, you can place it on hold, transfer it, set up a conference, enter optional account codes, etc., as described in this manual. The Model 8660 or 8662 has menu buttons for easy use of outside call features.

*If you dialed an invalid or restricted phone number*, you hear reorder tones and the display shows INVALID OUTSIDE NUMBER or RESTRICTED OUTSIDE NUMBER.
5. To use the speakerphone, if equipped: If off-hook, speak handsfree when the party answers. If on-hook, press the Speaker button and hang up.

   To change from speakerphone to handset: Lift the handset and speak. The Speaker button goes off. You may later transfer back to the speakerphone by pressing the Speaker button and hanging up.

6. To terminate the call: If on-hook, disconnect by pressing the Speaker button. If off-hook, hang up.

To place a call using Automatic Route Selection:

1. **Inter-Tel endpoints**: While on- or off-hook, enter the Automatic Route Selection feature code (92000), or press the ARS button. You hear dial tone.

   **Single line endpoints**: Lift the handset and enter the Automatic Route Selection feature code (92000). You hear a single progress tone followed by dial tone.

   If required, enter a forced account code using the dialpad or a pre-programmed Speed Dial button (see page 184). Display endpoints show ENTER ACCOUNT CODE and the digits as they are entered. You hear a progress tone when the code is accepted.

2. Dial the phone number. Use the MUTE button to backspace if you make a mistake. Include the area code, if needed. You hear one of the following signals.

   - **Single progress tone**: Your call is being placed. The associated individual trunk or Call button flashes slowly. Model 8660 or 8662 endpoint users have the same options as described in “Placing Outside Calls” on page 178.

   - **Busy signals, then music**: All of the trunks in the route group are busy. Do one of the following:

     1.) Camp-on and wait for the trunk.

     2.) Request a queue callback: The system places your endpoint in a queue for the trunk. When you hear ringing and the display shows TG XXXX <or group name> IS NOW AVAILABLE, lift the handset; Inter-Tel endpoint users may also need to press the fast flashing Call button or individual trunk button (or the ANSWER button). You hear dial tone and digits being dialed. The associated individual trunk or Call button flashes slowly.

   - **Repeating reorder tones**: You are restricted from the number dialed or from using the route selected. The display shows INVALID OUTSIDE NUMBER or RESTRICTED OUTSIDE NUMBER. If the number that you dialed does not match any of the programmed ARS route group patterns, the display shows UNABLE TO ROUTE CALL.
Receiving Outside Calls

To receive an outside call on an Inter-Tel endpoint:

You will hear one of the following signals when receiving an outside call.

- **Repeating long tones and a Call button or individual trunk button is flashing at the fast rate:** A call is ringing in. Lift the handset and/or press the flashing individual trunk button, flashing Call button, or \( \text{ANSWER} \) button. (Model 8660 or 8662 endpoint users who have a programmed Voice Mail extension can press the **SEND TO V-MAIL** menu button to transfer the call to Voice Mail.) The individual trunk or Call button flashes slowly during the call. Display endpoints show \(<\text{trunk name}>\) RINGING IN.

- **Intercom call or page announcing a call and a Call button or individual trunk button is flashing at the fast rate:** A call has been transferred to your endpoint. Lift the handset and/or press the flashing individual trunk button, flashing Call button, or \( \text{ANSWER} \) button. Model 8660 or 8662 endpoint users that have a programmed Voice Mail extension can press the **SEND TO V-MAIL** menu button to transfer the call to Voice Mail. The individual trunk or Call button flashes slowly during the call. Display endpoints show \(\text{TG} \, \text{XXXX} \) or \(<\text{group name}>\) TFER FROM EX \, \text{XXXX} \) or \(<\text{username}>\).

- **An individual trunk button is flashing at the fast rate, there is no ring signal, and you have allowed answer for the trunk. You may hear ringing on another endpoint:** Press the fast-flashing individual trunk button as described above or enter the Automatic Trunk Answer feature code (350). The \( \text{ANSWER} \) button cannot be used to answer these calls.

To receive a call on a single line endpoint:

- **You hear a page announcing a transfer and/or repeating long double tones:** A call is ringing in. Lift the handset; you are connected to the caller.

- **You hear a call ringing in on another endpoint and you have allowed answer:** Lift the handset and enter the Automatic Trunk Answer feature code (350). You are connected to the lowest-numbered trunk ringing in for which you have allowed answer.

To receive a call that was transferred to hold:

When you hear a page announcing that an outside call is holding at your endpoint (a Call button or individual trunk button is fluttering):

- **Inter-Tel endpoints:** Lift the handset and/or press the fluttering Call button or individual trunk button, or press the \( \text{ANSWER} \) button. You are connected to the caller.

- **Single line endpoints:** Lift the handset and press the Hold button or enter the Individual Hold feature code (336). You are connected to the caller.
On-Hook Monitoring

An Inter-Tel endpoint user with a speakerphone can monitor the call (listen to a recorded message, wait for the call to be answered, or wait on hold), and then speak handsfree when answered.

**To use on-hook monitor:**

1. While on a call, press the Speaker button and hang up. If you do not want to be heard and if the MUTE button is not lit, press the MUTE button or MUTE menu button, or press the Special button and enter the Microphone Mute On/Off feature code (314). You can hear the other party (if the microphone is muted, you cannot be heard). The Speaker button is lit (and the MUTE button is lit if the microphone was muted).

2. **To return to the conversation:** You may lift the handset or speak handsfree. (If the microphone was muted, press the lit MUTE button or the UNMUTE menu button to speak. The MUTE button goes off.)

Group Listen Feature

The Group Listen feature allows an Inter-Tel endpoint user to transmit a conversation over the endpoint speaker while in handset or headset mode. This allows other people in the room to listen to the conversation. However, the endpoint microphone remains disabled so that only the headset or handset user can speak.

This feature cannot be used on a handsfree call. The user must be on a call using the handset or a headset before entering the feature code. Group Listen cannot be used on single line endpoints.

**To turn the Group Listen feature on or off during a call:**

While on a call using the handset or a headset, press the Special button and enter the Group Listen feature code (312). You hear a confirmation tone, and the display shows GROUP LISTEN ON or GROUP LISTEN OFF. The other party will not hear the confirmation tone.

If using a headset when Group Listen is enabled, the volume control adjusts the volume of the headset, not the speaker.

When the Group Listen feature is active in handset mode, the Speaker button lamp will remain unlit. This allows the user to place the call into handsfree mode at any time during the call by pressing the Speaker button. When the feature is active in headset mode, the Speaker lamp is lit. Pressing the button will disconnect the call.

The Group Listen feature code can be programmed under a feature button just like any other feature. The button functions as a toggle. If the button has a lamp, it is lit when the feature is activated and unlit when the feature is turned off.

If the Group Listen feature is active when an OHVA call is received by the endpoint, the OHVA call will camp on because the speaker is busy.
Trunk Camp On and Busy Trunk Callback (Queue)

Camp On and Callback Queue requests allow the caller to wait for an available resource. A user waiting for a specific resource (a trunk or an endpoint) will be served before a user waiting for a group that contains that specific resource, such as a trunk group or hunt group. This includes camped on Emergency Call feature calls.

**Trunk Camp On**

When a user attempts to select a busy outgoing trunk or trunk group, the system sends a busy signal. The user can wait off-hook to camp on until the trunk is available.

A user can camp on to busy resources on other nodes and will hear the other node’s music-on-hold while camped on. When the system has several routing options for a call, it tries to use each route, in order, until it finds a free route. If it is unable to find a free route, the system tries once again to use the first route. If the first route is still busy, the system camps on to the first route.

**To camp on to an outgoing trunk group:**

1. When you hear a busy signal after selecting an outgoing trunk, do not hang up. You camp on and hear music until a trunk in the trunk group is available. The display shows WAITING FOR <trunk name>.

2. When you hear a single progress tone and outside dial tone, dial the phone number you want. The associated Call button or individual trunk button flashes slowly.

**Busy Trunk Callback (Queue)**

If you attempt to select a busy outgoing trunk group and hear a busy signal or if you camp on, you can request a queue callback and hang up until the system signals your endpoint that a trunk in the trunk group is available. Each endpoint can place only one queue request at a time. If a second request is made, the first request is canceled and replaced by the second request.

A user cannot queue onto busy resources on another node.

Queue callbacks must be answered before the Queue Callback timer expires. If a callback is not answered, the queue is canceled. If the endpoint is busy when a trunk in the queued trunk group becomes available, the queue request is placed at the end of the queue list.

**To queue a busy outgoing trunk:**

1. **Inter-Tel endpoints:** When you hear a busy signal while selecting a trunk. Or if you are camped on, press the QUEUE button, or press the Special button and enter the Queue Callback feature code (6) and hang up. The display shows QUEUE REGISTERED FOR <trunk name>.

   **Single line endpoints:** When you hear a busy signal while selecting a trunk or if you are camped on, hookflash and enter the Queue Callback feature code (6). Hang up.

2. Your endpoint rings when the queued trunk is available. Display endpoints show TG XXXX or <group name> IS NOW AVAILABLE.

3. **Inter-Tel endpoints:** Lift the handset and/or press the fast-flashing Call button, individual trunk button, or the ANSWER button. Then dial the phone number. The associated Call button or individual trunk button flashes slowly. If ARS was used when the call was placed originally, the number is dialed automatically.

   **Single line endpoints:** Lift the handset and dial the number you want. If all DTMF decoders are busy when your endpoint is called back, the system sends repeating reorder tones instead of dial tone, and the queue is canceled. If ARS was used when the call was placed originally, the number is dialed automatically.
To cancel a queue before the callback:

**Inter-Tel endpoints:** While on-hook, press the **CANCEL QUEUE** button or **QUEUE** button. Or, enter the Queue Callback feature code (6). The display shows **QUEUE REQUEST CANCELED**.

**Single line endpoints:** Lift the handset and enter the cancel Queue Callback feature code (6). Hang up.

### Account Codes

Account codes are 3- to 12-digit codes that can be used in conjunction with the Station Message Detail Recording (SMDR) feature to aid record keeping. Account codes can be assigned to measure endpoint use and/or to identify calls for customer billing. There are Standard, Forced, and Optional types of account codes. The account code, when used, is recorded in the SMDR report as soon as the call is completed. If more than one account code is entered during a call, the last account code that was entered is recorded.

### Programmed Account Codes

Any endpoint can be assigned a standard account code or one of six types of forced account codes (four are ARS-dependent). Or, if you want, the endpoint can lack an associated account code.

There can be up to 512 standard and 512 forced account codes in the database.

The account code types are:

- **Standard**: Standard account codes automatically appear in the SMDR printout to identify calls from this endpoint. The user is not required to enter any digits during the call.

- **Forced – All Calls Validated**: If this type of account code is enabled, the user must enter a forced account code before an outside call can be placed. If the code matches one of the pre-programmed forced account codes in the database, the call is allowed. If the code does not match, the call is blocked. Endpoints with this account code type cannot forward calls to outside phone numbers.

- **Forced – All Calls Non-Validated**: With this account code type, the user must enter an account code before placing an outside call. The code is not checked against any lists, and the call is allowed as soon as the code is entered. Endpoints with this account code type cannot forward calls to outside phone numbers.

- **Forced – Local Toll Calls Validated**: This account code type requires the user to enter an account code after dialing a local toll call number (a toll call within the local area codes) or a long distance call when using ARS. If the code matches one of the pre-programmed forced account codes in the database, the call is allowed. If the code does not match, the call is blocked. Endpoints with this account code type cannot forward calls to outside phone numbers.

- **Forced – Local Toll Calls Non-Validated**: This account code type requires the user to enter an account code after dialing a local toll call number (a toll call within the local area codes) or a long distance call when using ARS. The code is not checked against any lists, and the call is allowed as soon as the code is entered. Endpoints with this account code type cannot forward calls to outside phone numbers.

- **Forced – Long-Distance Toll Calls Validated**: When this account code type is enabled, the user only has to enter an account code if the system detects that a long distance call (a toll call outside of the local area codes) has been dialed when ARS is used. If the code matches one of the pre-programmed forced account codes in the database, the call is allowed. If the code does not match, the call is blocked. Endpoints with this account code type cannot forward calls to outside phone numbers.
• **Forced – Long-Distance Toll Calls Non-Validated:** When this account code type is enabled, the user only has to enter an account code if the system detects that a long distance call (a toll call outside of the local area codes) has been dialed when the ARS is used. The code is not checked against any lists, and the call is allowed as soon as the code is entered. *Endpoints with this account code type cannot forward calls to outside phone numbers.*

• **None:** If the user will not be required to enter an account code, this option is selected. The user may still use optional account codes or enable the Account Code For All Calls Following feature, as appropriate.

If the system is installed in a network, the network validates account codes against the account code table on the same node as the caller. The account code follows the call as it moves from node to node and appears on every SMDR record associated with the call. However, if an optional account code is entered during the call, that account code will be used in the SMDR report.

If ARS Dial Rule #5 (Add Account Code) is enabled, the account code must be entered before the end of dialing for it to be included in the dialed digits. Forced, standard, and “all calls following” account codes are automatically entered before the digits. If using an optional account code, it must be entered after seizing a trunk, but before dialing the number, to be included in the ARS digits.

**Optional Account Codes**

Optional account codes can be entered at any time during a call. These user-defined codes are not preprogrammed, but must be within the maximum length set in programming. If entered, optional account codes are printed in the SMDR report for that call in place of standard or forced account codes that may have been used.

**To enter an optional account code after selecting a trunk or while on an outside call:**

1. **Inter-Tel endpoints:** Press the Special button and enter the Optional Account Code feature code (390). Display endpoints show ENTER ACCT CODE. The outside party, if any, does not hear you enter the code.

   **Single line endpoints:** Hookflash and enter the Optional Account Code feature code (390). The outside party, if any, hears music until step 2 is completed.

2. Enter the account code using the dialpad; Inter-Tel endpoint users may press a Speed Dial button to enter an account code number that has been stored in that Speed Dial number location. (Display endpoints show numbers. You hear a single progress tone when the code is accepted. Single line users are reconnected when the system accepts the account code. (If an account code is entered that has fewer than the programmed maximum number of digits, press # to return to the call. Pressing # will cancel the account code entry.)
Enabling the Account Codes for All Calls Following Feature

You can use the Account Code For All Calls Following feature to enter an account code once and apply it to all calls placed from that endpoint until the feature is canceled. The account code is stored in system memory and is used for all calls made by that endpoint. It affects other account code programming as follows:

- **Forced**: If forced account codes are enabled for the endpoint, the “all calls following” account code is used for all calls and the user will not be prompted to enter an account code until the “all calls following” code is canceled.

- **Standard**: If standard account codes are enabled for the endpoint, the “all calls following” account code overrides the standard code.

- **None**: If no forced or standard account codes are enabled for the endpoint, the “all calls following” code will be used as an optional account code.

- **Optional**: In the event that this feature is in effect and an optional account code is also entered, the optional account code will override the “all calls following” account code for that call only. All subsequent calls will be associated with the “all calls following” account code.

To cancel this feature, the feature code is entered without an account code (just press to terminate programming).

**To enter an account code and use it for all following calls:**

1. **Inter-Tel endpoints**: Enter the Account Code For All Calls Following feature code (391). Display endpoints show ENTER ACCT CODE.

   **Single line endpoints**: Lift the handset and enter the Account Code For All Calls Following feature code (391).

2. Enter the account code using the dialpad; Inter-Tel endpoint users may press a Speed Dial button to enter an account code number that has been stored in that Speed Dial number location. If an account code is entered that has fewer than the programmed maximum number of digits, press to terminate the code. Display endpoints show numbers. You hear a single progress tone when the code is accepted.

**To cancel the account code for all calls following feature:**

1. **Inter-Tel endpoints**: Enter the Account Code For All Calls Following feature code (391). Display endpoints show ENTER ACCT CODE.

   **Single line endpoints**: Lift the handset and enter the Account Code For All Calls Following feature code (391).

2. **If on-hook**, lift and replace the handset, or press to exit. **If off-hook**, hang up.
Placing Calls On Hold

There are several ways to place intercom and outside calls on hold. While on hold, the caller hears music (if equipped). There are three hold applications in the system:

- **Individual Hold** places the call on hold at one endpoint. It can then be picked up directly at that endpoint or it can be picked up at another endpoint using the Reverse Transfer feature.

- **System Hold** places the call on hold so that it can then be picked up directly at any Inter-Tel endpoint that has an individual trunk button and has allowed-answer and/or outgoing access for the associated trunk or at the endpoint that placed it on hold. Attempting to place a conference on system hold will place the conference on individual hold. Intercom calls cannot be placed on system hold. Single line endpoints cannot place calls on system hold; attempting to do so at a single line endpoint will place the call on individual hold. In a network setting, when a user puts a trunk on system hold, only that user and users on the same node as the trunk can access the call.

- **Consultation Hold** allows a single line endpoint user to pause during a call, use other system features, and then return to the caller by hookflashing. If the user attempts to hang up after placing a call on consultation hold, the call recalls the endpoint.

If a call remains on hold until the Hold timer expires, it recalls the endpoint where it is on hold, and the Recall timer is started. If it is still unanswered when the Recall timer expires, it recalls the endpoint’s attendant, and the Abandoned Call timer is started. If the endpoint does not have an attendant, the call continues to recall at the endpoint that placed it on hold. If the call is not answered before the Abandoned Call timer expires, the call is disconnected by the system.

For users’ convenience, the system has two Hold timers: **Hold** and **Hold–Alternate**. In the default state, the Alternate timer is set for a longer time period than the Hold timer. However, both timers are programmable. The “Alternate Hold Timer” endpoint flag determines which timer each endpoint will use. If the flag is disabled, the endpoint uses the Hold timer. If it is enabled, the endpoint uses the Alternate timer.

Inter-Tel endpoint users can avoid the Hold timer by muting the microphone during a call instead of placing the call on hold. If this is done, the caller will not hear Music-On-Hold and will not recall the endpoint.

**Individual Hold**

To place an intercom or outside call on individual hold:

1. While on the call:
   
   **Inter-Tel endpoints**: Press the Hold button or press the Special button and enter the Individual Hold feature code (336). You hear intercom dial tone and the Call button, individual trunk button, or \[IC\] button flutters.

   **Single line endpoints**: Hookflash and enter the Individual Hold feature code (336). You hear intercom dial tone.

2. Hang up or place another call.

To return to the caller on individual hold:

**Inter-Tel endpoints**: Lift the handset and/or press the fluttering \[IC\] button for intercom calls, or the Call button or individual trunk button (or the \[ANSWER\] button) for outside calls.

**Single line endpoints**: Lift the handset and enter Individual Hold feature code (336).
System Hold

This feature does not apply to single line endpoints.

To place an outside call on system hold (Inter-Tel endpoints only):

1. While on a call, press the Special button and enter the System Hold feature code (335). You hear intercom dial tone, and the associated Call button or individual trunk button flutters.
2. Hang up or place another call.

To return to a call on system hold (Inter-Tel endpoints only):

At any Inter-Tel endpoint that shows a fluttering individual trunk button for the call and has allowed-answer and/or outgoing access for that trunk, lift the handset and/or press the fluttering button.

At the Inter-Tel endpoint that placed the call on hold, press the fluttering individual trunk or Call button.

Intercom calls and conferences cannot be placed on system hold. If you attempt to do so, the call or conference will actually be placed on individual hold.

Consultation Hold

To use Consultation Hold (single line endpoints only):

1. While on a call, hookflash to place the call on consultation hold. You hear three short tones followed by dial tone.
2. You can place a call or enter a feature code.
   If you hang up while the call is on consultation hold, the call immediately recalls your endpoint. If you hang up after dialing an extension number, the call transfers to that endpoint.
3. Return to the caller on hold by hookflashing.

Hold Recall

To answer a hold recall:

When the recall rings at your endpoint, the display shows <endpoint or trunk> RECALL-ING. Lift the handset or press the ANSWER button. Executive Display, Professional Display, and Model 8560 endpoint users can press the ANSWER menu button. Single line endpoints are automatically connected; Inter-Tel endpoint users may need to press the flashing Call or individual trunk button.
Microphone Mute

Whether handsfree or using the handset, you can temporarily turn off your microphone while on a call. The call is still connected; you can hear the other party, but they cannot hear you. Since the call is not placed on hold, no timer is activated. The MUTE button is lit when the microphone is muted; the light will go off when you press the MUTE button to re-enable the microphone.

If you receive an Off-Hook Voice Announce call (see page 176), you can press the MUTE button while speaking to the OHVA caller to mute the handset microphone. The caller on the handset will not hear you or the OHVA caller. When you press MUTE again, the handset microphone is re-enabled.

You cannot mute the microphone on your endpoint while your call is on hold at another endpoint.

If you receive complaints from IP endpoint users about losing audio on muted calls, there may be a firewall issue. Contact your local Inter-Tel dealer for assistance.

Muting the endpoint microphone:

1. To temporarily turn off your microphone while on a call, press the MUTE button or MUTE menu button, or press the Special button and enter the Microphone Mute On/Off feature code (314). The MUTE button is lit, and the display shows MICROPHONE MUTE ON. If off-hook, do not hang up (unless you press the Speaker button first).

2. Press the lit MUTE button or UNMUTE menu button, or press the Special button and enter the feature code (314) to turn the microphone on. The MUTE button goes off, and the display shows MICROPHONE MUTE OFF.

Call Waiting

While an endpoint is in use, incoming intercom and outside calls camp on until the busy endpoint is available. The busy party hears a single Camp On tone every 15 seconds, unless the Camp On Tone timer is changed or Camp On tones are disabled at the endpoint.

An endpoint programming flag titled Camp On Indications allows Camp On tones to be enabled or disabled for incoming intercom and outside calls. This flag is programmed on an endpoint-by-endpoint basis.

The DID/E&M Receive Busy Instead Of Camp-On endpoint flag determines whether E&M and DID callers will receive busy signal or receive ringback and camp on when calling a busy endpoint. In the default state, busy tones are disabled, and the callers will hear ringback while camped on to the called endpoint. This flag is programmed on an endpoint-by-endpoint basis.

To respond to call waiting using an Inter-Tel endpoint:

1. If, while on a call, you hear a single camp-on tone and/or see the IC button, a Call button, or an individual trunk button flashing rapidly, you have a call waiting. The display indicates the type of call that is waiting (intercom, outside, recall, etc.). If you are a hunt group member, the display shows the number of calls camped on to the hunt group. Do one of the following:
   - If you want to end the current call, hang up. A waiting outside call rings as an incoming call; an intercom call rings as a private call. Answer as usual.
   - If you want to place the current call on hold, press the Hold button or press the Special button and enter the Individual Hold feature code (336). The Call button, individual trunk button, or IC button flutters (unless an intercom call is
camped on, in which case the IC button flashes rapidly until the camped on call is picked up, then the IC button flutters). Answer the waiting call by pressing the flashing Call button, individual trunk button, or IC button.

- **If using Model 8660 or 8662**: You may press one of the following menu buttons:
  - **ANSWER**: Press this button to end the current call and answer the ringing call.
  - **SEND TO V-MAIL**: This button appears only if you have a Voice Mailbox. Press this button to transfer the call to your mailbox.
  - **IC CALL TO DND**: Pressing this button will send Do-Not-Disturb tones to the caller. If using a display Inter-Tel endpoint, the caller will see **DO-NOT-DISTURB**.

2. **To return to the holding call**: Press the fluttering individual trunk button, Call button, or IC button.

**To respond to call waiting using a single line endpoint**:

When you hear a single camp-on tone every 15 seconds while you are on a call, another call is waiting. Do one of the following.

1. **If you want to end the current call**, hang up. The call rings at your endpoint. Lift the handset to speak to the waiting caller.

   **If you want to place the current call on hold**, hookflash and then enter the Individual Hold feature code (336). You are connected with the waiting caller.

2. **To return to the first call on hold**:

   **If you want to end the current call**, hang up. Return to the holding call by lifting the handset and entering the Individual Hold feature code (336).

   **If you want to place the current call on hold and connect with the original call**: Hookflash and then enter the Individual Hold feature code (336) twice.

**To move a camped-on intercom call to Do-Not-Disturb**:

While on a call, you hear call waiting signals:

- **Endpoints with menu buttons**: Press the **DO-NOT-DISTURB**, or **IC CALL TO DND** menu button.

- **Other Inter-Tel endpoints**: Press the DND button or enter a Do-Not-Disturb feature code (370 or 372). The DND button lights. The intercom caller is no longer camped on and receives Do-Not-Disturb indications.

- **Single line endpoints**: Hookflash and enter a Do-Not-Disturb feature code (370 or 372). The intercom caller is no longer camped on and receives Do-Not-Disturb indications. Hookflash twice to return to the current call.
Call Transfer

The following paragraphs describe transfer feature codes and the different ways to transfer calls.

Feature Codes

Either of two feature codes can be used to transfer intercom and outside calls to other endpoints, voice processing system applications, hunt groups, off-node devices, or outside phone numbers. The call transfer options are as follows:

- **Transfer to ring**: You can transfer intercom or outside calls to another endpoint, a voice processing system application, a hunt group, or an outside phone number.
- **Transfer to hold**: Either intercom or outside calls can be transferred to another endpoint and placed on hold using this feature.

Transferring Conference Calls

An established conference can be transferred to an endpoint. While the transfer is taking place, the parties in the conference remain connected to each other and may converse. The transfer will appear at the destination endpoint in the same manner as any other transferred call along with a CONFERENCE TFR FROM <username> display and may be answered by the endpoint user.

Transfer to a System Forward

When a user transfers a call to the principal station of a system forwarding path, the Handsfree Announce System Forward system-wide flag determines what the caller will hear. If enabled, the caller will place a handsfree call to the principal station and can announce the transfer. The call does not enter the forwarding path. If the option is disabled, the transferring user will hear ringing and can announce the call only if it is answered by the principal station or one of the forwarding points.

Transfers and Outside Calls

If a user answers another call while a party is waiting on transfer hold, the waiting party will be moved to individual hold at that endpoint. If the Hold timer expires while a call is on transfer hold, the call will recall the endpoint. The user must return to the call on hold and begin the transfer process again. There may be some reduction in voice volume when an outside call is transferred to an outside phone number depending on the quality of the trunks in use.

Transfer-to-Connect Endpoint Flag

If the Transfer-To-Connect Allowed endpoint flag is enabled, an announced transfer call (including an announced transfer from Voice Mail) connects immediately with the destination once the transferring party hangs up. (If the flag is disabled, the recipient must press a Call button or the ANSWER button to answer the transferred call). If the user has a headset enabled, or if the user picks up the handset for the initial call and the transferring party disconnects, the transferred party hears a single tone and then is immediately connected without any further action from the user. Calls will not automatically connect in handsfree speakerphone mode. This flag applies only to Inter-Tel endpoints and is disabled in the default state.
Transfer to Ring

To transfer a call or conference:

1. **Inter-Tel endpoints**: While on a call, press the `TRANSFER` button, or press the Special button and enter the Transfer to Ring feature code (345). Intercom dial tone is heard. The display shows TRANSFER CALL TO EXTENSION.

   **Single line endpoints**: While on a call, hook flash. Tones are heard, followed by intercom dial tone.

2. Perform one of the following actions:
   - Select an outgoing trunk group or press the OUTSIDE PHONE menu button, and dial the phone number you want. *If all trunks in the trunk group are busy*, select another trunk group or camp on and wait for a trunk. *If you hear repeating reorder tones*, you dialed a restricted phone number. To return to the caller on transfer hold, press the lit Call button, trunk button, or IC button (Inter-Tel endpoint) or hookflash twice (single line endpoint).
   - Dial the extension number you want or press a Speed Dial button, the VOICE MAIL menu button, or the MESSAGE CENTER menu button. Model 8660 or 8662 users can use the Directory feature by pressing the IC DIRECTORY menu button. The number can be an endpoint extension number, for the attendant, a hunt group pilot number, or a Voice Mail extension number.

   *If you transfer to an endpoint that is forwarded to Voice Mail*, you hear repeating double tones and the display shows DEST FORWARDED TO VOICE MAIL.

   *If you transfer to Voice Mail*, the system waits for you to enter the mailbox number. The display shows ENTER MAILBOX. Enter the mailbox number you want. Users of Executive and Professional Display endpoints and Model 8560 endpoints can access the Directory feature by pressing the IC DIRECTORY menu button, if the system is validating mailbox numbers. *If you do not enter a mailbox number before you hang up*, the caller is connected to the Voice Mail unit and must enter the mailbox number after listening to the introductory voice prompts.

3. If appropriate, wait for an answer and announce the call. If you are calling an endpoint such as Automated Attendant that does not accept call announcements, the display shows HANG UP TO FINISH TRANSFER.

4. You have the following options:
   - **To complete the transfer**: Hang up or press another Call button. The display shows CALL TRANSFERRED TO <user name>.
   - **To return to the caller**: Press the fluttering Call button, trunk button, or IC button (Inter-Tel endpoint) or hookflash twice (single line endpoint).
   - **To try another endpoint**: Press the `TRANSFER` button or press the Special button and enter the Transfer to Ring feature code (345). Then dial another extension number.
   - **To transfer the call to hold when calling an endpoint (Inter-Tel endpoints only)**: Press the HOLD button or the TRANSFER-TO-HOLD menu button. Or press the Special button and enter the Individual Hold feature code (336). The call will not ring or send Call Waiting signals until the Hold timer expires.
Transfer to Hold

A call transferred to hold at an endpoint does not ring or send a display message while holding. After the Hold timer expires, the endpoint rings or sends Call Waiting signals. Also, calls transferred to Hold do not recall to the transferring party’s endpoint. Rather, they recall to the receiving party’s attendant if unanswered after the Hold and Recall timers expire. Besides the procedure listed below, Inter-Tel endpoint users have an alternative method for transferring calls to hold; see step 4 in the preceding procedure.

To transfer a call to Hold:

1. **Model 8660 or 8662**: While on a call, press the **TRANSFER-TO-HOLD** menu button. The display shows **TRANSFER CALL TO EXTENSION**.

2. **Other Inter-Tel endpoints**: While on a call, press the Special button and enter the Transfer To Hold feature code (346). The display shows **TRANSFER CALL TO EXTENSION**.

3. **Single line endpoints**: While on a call, hookflash and enter the Transfer To Hold feature code (346).

2. Model 8660 or 8662 users can use the Directory feature by pressing the **IC DIRECTORY** menu button.

3. Hang up to complete the transfer.

To pick up a call that was transferred to Hold:

If a call was transferred to Hold at your endpoint:

1. **Inter-Tel endpoints**: When your **IC** button, a Call button, or individual trunk button is fluttering, lift the handset and/or press the fluttering button or the **ANSWER** button for an outside call. You are connected to the caller. You can also connect to the call by lifting the handset and dialing the Individual Hold feature code (336).

2. **Single line endpoints**: Lift the handset and enter the Individual Hold feature code (336). You are connected to the call.

Transfer Recalls

If a call is transferred to another endpoint and is not answered before the appropriate Transfer timer expires (and if it was not transferred to hold), the call recalls the transferring endpoint’s recall destination (defaults to the endpoint) and the Recall timer is started. The call rings until the Recall timer expires. If unanswered, it recalls the transferring party’s attendant, and the Abandoned Call timer is started. (If the transferring endpoint has no attendant, the call continues to recall at the transferring endpoint.) If the call is not answered before the Abandoned Call timer expires, the call is disconnected by the system.

In a network, an endpoint’s transfer recall destination can be an off-node device.

To answer a transfer recall:

1. **Inter-Tel endpoints**: When the recall rings at your endpoint, the display shows **<username> RECALLING FROM <username>**. Lift the handset and/or press the **ANSWER** button. Press the medium-flashing Call button or individual trunk button, or press the **ANSWER** button if the call is not automatically connected.

2. **Single line endpoints**: If you hear a repeating signal of four tones and a pause, lift the handset. You are immediately connected.
Call Screening

Calls transferred from the Automated Attendant or a Call Routing Announcement application can be screened, announced, or unannounced. Separate programming flags in the database determine the methods used for transferring calls to endpoints with mailboxes, endpoints with extension IDs, and extensions without mailboxes or IDs. For details, see “Call Screening Options” on page 268.

Reverse Transfer

Calls ringing or recalling at an endpoint or a hunt group can be picked up at any other endpoint, using this feature. Calls on hold can also be picked up from endpoints. A user can reverse transfer a call from an endpoint or a hunt group on another node.

If more than one call is ringing or holding at the endpoint or hunt group, a priority list determines which call is reverse transferred. Calls are selected in the following order, and if more than one call of the same type is at the endpoint, the calls are picked up in the order they were received:

1. Ringing calls
2. Camped-on calls
3. Holding outside calls
4. Holding intercom calls

If a call is reverse transferred from a hunt group announcement or overflow endpoint, and an endpoint in the hunt group becomes available, the call will be disconnected from your endpoint immediately when answered by the hunt group endpoint.

Group Call Pick-up: When the Group Call Pick-Up feature is enabled, a call ringing into a hunt group or one of its endpoints can be picked up at any other endpoint. Users can enter the Reverse Transfer feature code (4) and dial a hunt group’s pilot number to pick up a call that is ringing into the hunt group’s pilot number or to any endpoint within that hunt group. See page 132 for details about the Group Call Pick-Up feature.

Group Call Pick-up does not affect DSS lamp status any differently from standard hand group calls or individual calls.

There is a programmable endpoint flag called “Transient Call Indication On Call Answer” that determines whether the user sees a call display when reverse transferring a call (or using a secondary extension button). If the flag is enabled, the display indicates whether the call was ringing, recalling, transferred, or holding at the other endpoint.

**To reverse transfer a call, take one of the following actions:**

- Lift the handset (Inter-Tel endpoints can remain on-hook) and enter the Reverse Transfer (Call Pick-Up) feature code (4), then dial the extension or pilot number where the call is ringing or holding. You are automatically connected to the caller.

- Lift the handset (Inter-Tel endpoints can remain on-hook) and dial the extension or pilot number where the call is ringing or holding, then press the Special button or hookflash and enter the Reverse Transfer (Call Pick-Up) feature code (4). You are automatically connected to the caller.
Conference Calls

Users can establish multi-party conference calls without operator assistance. There can be up to four parties in a conference. In addition to the initiating endpoint, the conference can include any combination of up to three intercom and/or outside calls.

During a conference call, some reduction in voice volume may be noticed, depending on CO trunk quality.

During a conference, if any user presses a dialpad button, the DTMF tones will be heard by all other parties in the conference. This allows conference callers access to DTMF-controlled devices.

Using the call transfer feature described on page 191, an established conference can be transferred to another endpoint. While the transfer is taking place, the parties in the conference remain connected to each other and may converse. The transfer will appear at the destination endpoint in the same manner as any other transferred call along with a CONFERENCE TFR FROM <user name> display and may be answered by the party.

NOTICE

While Inter-Tel systems are designed to be reasonably secure against CO trunk misuse by outside callers, there is no implied warranty that the Inter-Tel system is invulnerable to all forms of unauthorized use.

For example, unauthorized use could occur if the CO/local exchange does not provide “supervision” on the telephone lines connected to the system. Without CO/local exchange line supervision a party might not be disconnected from the Inter-Tel system when they hang up. Without line supervision it is possible for the caller to remain connected to a CO/local exchange trunk circuit that is connected to the Inter-Tel system. If the caller then begins dialing, the call could be placed through the Inter-Tel system and would then be billed to the Inter-Tel system’s owner.

The Inter-Tel system cannot check or control this type of call for toll restriction and may not register the call in SMDR. This situation could occur when a call is connected to an endpoint on a line without supervision or when a conference call is placed on a line without supervision.

In a network setting, a user can build a conference with party members (endpoints and/or trunks) on other nodes. The conference circuits used to build the conference will be allocated from the originating party’s node. Any party, not just the originating party, can add a party member to the conference.

- If the party who adds the new member is on the originating party’s node, the party will be added to the conference circuit already allocated. This conference is still restricted to the four party limit.
- If the party who adds the new member is not on the originating party’s node, a new conference circuit is allocated from the second node. Because this allows more than four parties in a conference, users should be warned that voice levels can be degraded as more parties are added to the conference.
Placing a Conference Call

To place a conference call while on a call with an Inter-Tel endpoint:

Press the CNF button, or press the Special button and enter the Conference feature code (5). The party is on conference wait hold (hears music) and the CNF button flutters. The display shows CALL NEXT PARTY TO CNF.

To place a conference call while on a call with a single line endpoint:

1. Hookflash and enter the Conference feature code (5). The party is on conference wait hold (hears music).
2. Place an intercom or outside call to another party (or return to a call on hold) to be included in the conference. Users of Executive and Professional Display endpoints and Model 8560 endpoints can press the IC DIRECTORY or OUTSIDE DIR menu buttons to look up numbers. An existing conference is not a valid party for another conference.

To place the party on conference wait hold with an Inter-Tel endpoint:

Press the CNF button again. Or, press the Special button and enter 5 again.

To place the party on conference wait hold with a single line endpoint:

Hookflash and enter the Conference feature code (5). The party is on conference wait hold (hears music).

3. Repeat step 2 until all parties are on conference wait hold (or the maximum allowed number has been reached).
4. Connect all waiting parties in the conference:

   To connect all waiting parties in the conference with an endpoint that has menu buttons:

   Press the CONNECT TO CONFERENCE menu button. All parties are connected. The display shows CNF IN PROGRESS. The CNF button and a Call button flash slowly. Use the following menu buttons:

   HOLD: Press this button to place the conference on hold.

   DISCONNECT: Press this button to remove your endpoint from the conference. If there are no other endpoints in the conference, all parties are disconnected. If there is still another endpoint in the conference, the parties remain connected.

   ADD PARTY: This button appears only if you have not reached the maximum number of allowed parties. Press this button to leave the conference parties connected while you place another call to be added to the conference.

   MUTE: Press this button to switch the microphone on or off during the conference.

   TRANSFER CNF: Press this button and enter an extension number to transfer the conference to another endpoint or the Voice Mail application.

   To connect all waiting parties in the conference with another type of Inter-Tel endpoint:

   Press the CNF button again. Or, press the Special button and enter 5 again. All parties are connected. Display endpoints show CNF IN PROGRESS. The CNF button and a Call button flash slowly.
To connect all waiting parties in the conference with a single line endpoint: When all parties are on conference hold, hookflash [recall], and enter the Conference feature code (5). All parties are connected.

If the number of parties in your conference exceeds the capacities of the system resources, you will hear reorder tone and on display endpoints see MAX NUMBER OF PARTIES EXCEEDED. To complete the conference setup, return to the parties on conference hold. Release parties one-by-one until fewer than the maximum remain on the conference. To release parties from Inter-Tel endpoints, press the associated Call or IC button; from single line endpoints, enter the Individual Hold feature code (5) twice. When the maximum number is no longer exceeded, repeat step 4 above to connect the conference.

Add Parties to a Conference

When on a conference call, parties cannot be added when the maximum number of participants has been reached.

To add parties to an existing conference (any inside party) on an Inter-Tel endpoint:

1. During the conference, press the button. The other parties remain connected. The Call button and CNF button flutter.
2. Place an intercom or outside call to the other party (or return to a call on hold) to be included in the conference and press the CNF button.
3. To connect all parties in a conference: Press the CNF button again.
4. To add another party: Repeat steps 1-3 again.

To add parties to an existing conference (any inside party) on a single line endpoint:

1. During the conference, hookflash. The other parties remain connected.
2. Place an intercom or outside call to the other party to be included in the conference (or return to a call on hold). Hookflash and enter the Conference feature code (5).
3. To connect all parties in a conference: Hookflash and enter the Conference feature code (5) again.
4. To add another party: Repeat steps 1–3.

Exiting a Conference

There are several options for leaving the conference:

- Exit the conference by hanging up. The other parties remain connected only if at least one is the endpoint of an inside party. You cannot return to the conference. If there are no inside party endpoints remaining in the conference, it is disconnected.
- Leave the conference and allow the outside parties to remain connected by pressing the CNF button or entering the Conference feature code. You can return to the conference at any time.
- End the conference and place all of the parties on individual hold. You can return to the parties one at a time.
- Place the entire conference on hold. The conferenced parties remain connected.

If a user attempts to place a conference call on system hold, the conference will be placed on individual hold.
Exiting and Re-Entering the Conference

To exit the conference and leave the other parties connected (any inside party) with an Inter-Tel endpoint:

During the conference, press the CNF button and hang up. The CNF button and a Call button flutters.

If all remaining parties are outside parties and the Hold timer expires while you are out of the conference, the conference recalls your endpoint. The recall rings at your endpoint and the CNF button flashes at the medium rate. The display shows CONFERENCE RECALLING. Press the CNF button to return to the conference. If the recall is not answered before the Recall timer expires, it will recall the attendant.

To return to the conference:

Lift the handset and/or press the CNF button again.

To exit the conference and leave the other parties connected (any inside party) with a single line endpoint:

During the conference, hookflash and hang up. If the Hold timer expires while you are out of the conference, the parties recall your endpoint. When the recall rings at your endpoint, lift the handset to answer the recall. If the recall is not answered before the Recall timer expires, it will recall the attendant.

To return to the conference:

Lift the handset and/or enter the Conference feature code (5).

Ending the Conference and Placing All Parties on Hold

To end the conference and put all parties on hold (any inside party) with an Inter-Tel endpoint:

During the conference, press the CNF button and hang up. The CNF button and a Call button flutters. The parties are still connected in a conference.

To place the parties on individual hold: Press the Hold button or enter the Individual Hold feature code (336). You can then return to each party individually by pressing the associated Call or IC button.

To end the conference and put all parties on hold (any inside party) with a single line endpoint:

During the conference, hookflash and hang up. The parties are still connected in a conference.

To place the parties on individual hold: Lift the handset and enter the Individual Hold feature code (336). You can then return to each party individually by entering the Individual Hold feature code again.
**Placing the Entire Conference on Hold**

Any inside party involved in a conference can place the conference on hold at his or her endpoint. The other conference parties remain connected and can converse. The user that placed the conference on hold can then place calls or use system features while the conference is on hold. Then the user can return to the conference on hold. Note, however, that a conference may not be terminated while it is on hold.

After a user has moved a conference to hold, the user may, if appropriate, establish another conference. Because each conference is established under a separate Call button, it is possible for the user to have access to any individual conference for further call processing. If the user elects to use the  button for conference access, the  button will rotate through the conferences in priority order. The  button will always reflect the status of the top-priority conference. The priority order for the  button when multiple conferences are in effect is:

- Conference or call on transfer hold
- Ringing (ring in, recall, or camp on) conference
- Conference on hold
- Active conference

**To place the conference on hold (any inside party) during a conference call with an Inter-Tel endpoint:**

Press the Hold button or press the Special button and enter the Individual Hold feature code (336). The conferenced parties remain connected. The call and  buttons flutter.

**To place the conference on hold (any inside party) during a conference call with a single line endpoint:**

Hookflash and then enter the Individual Hold feature code (336). The conferenced parties remain connected.

If you do not return to the conference before the Hold timer expires, the conference will recall your endpoint. If the recall is not answered, the conference will recall to your attendant's endpoint. At the attendant, the call will be terminated automatically after the Abandoned Recall timer if no inside parties are still in the conference and if it is not answered by the attendant. (If the attendant answers the call, a conference is re-established with the attendant endpoint as one of the conference parties.)

**To return to the conference with an Inter-Tel endpoint:**

Press the Hold button, the  button, or the fluttering Call button.

**To return to the conference with a single line endpoint:**

Lift the handset and enter the Conference feature code (5).
Record-A-Call

If your system is programmed with a Record-A-Call application, the endpoints can be programmed to use the Record-A-Call feature. It allows users to enter a feature code whenever they want to record an ongoing call in their designated Record-A-Call mailbox. Users can retrieve the recorded messages later, just as they would any other mailbox messages.

When a user requests the Record-A-Call feature, the system establishes a conference call with the current call parties and a mailbox. If there are no conference circuits available when the user requests Record-A-Call, or if there are already four parties on the call, the user hears reorder tones and cannot use the feature.

The Record-A-Call mailbox records the call as a Voice Mail message. All parties will be included in the recording. The Record-A-Call mailbox can be programmed to play a message announcing that the Record-A-Call feature is in progress. There can be separate messages for day and night modes.

To signal to the other parties that the Record-A-Call feature is in use, a system-wide Record-A-Call Tone flag can be enabled during system programming. If the flag is disabled, there will be no alerting tone. If enabled, the beep will occur periodically throughout the call. The Record-A-Call Tone Interval timer determines how often this tone is generated. If the timer is set to 0, the tone is generated only when the feature is first activated.

There are two ways an endpoint can be programmed to use this feature:

- The endpoint can be programmed to use its personal mailbox, or another mailbox, as the assigned Record-A-Call mailbox. Only this assigned mailbox can be selected. This state is the default programming for all endpoints.
- The endpoint can be programmed with a default mailbox, but with the option of selecting a different mailbox. If the user chooses to not enter a mailbox number, the system automatically selects the default mailbox.

In a network setting, the Record-A-Call destination does not have to be on the same node as the endpoint, but the voice processing system must be programmed with a mailbox for that endpoint.

When the requesting user hangs up, all parties are automatically disconnected. If all parties on the call hang up except the requesting user, an intercom call will remain connected between the requesting endpoint and the Record-A-Call application. This allows the user to make additional comments before ending the recording.

When the user turns off Record-A-Call or ends the call being recorded, the system delivers the message to the mailbox. The associated endpoint will receive Message Waiting indications as usual.

A monitored endpoint can initiate a Record-A-Call session without terminating call monitoring. Hunt group supervisors, however, cannot initiate a Record-A-Call while monitoring an endpoint, and an endpoint using Record-A-Call cannot be monitored. In other words, to use Record-A-Call and station monitoring simultaneously, the supervisor must first be monitoring the endpoint, and the endpoint, not the supervisor, must initiate the recording.

The Record-A-Call feature code can be assigned to a user-programmable feature button. If the feature button has a lamp, it flashes while the Record-A-Call conference is being set up and is lit while the feature is active.
The programmer can set a maximum length for Record-A-Call messages. The Record-A-Call Maximum Message Length timer can be set at 0–600 minutes. A 0 setting allows messages of any length, limited only by the available disk space. The default is 30 minutes. (This overrides the mailbox’s maximum message length setting. However, if a Record-A-Call message exceeds the maximum message length, the mailbox will be considered full until that Record-A-Call message is deleted.)

As with any other conference call, any inside party involved in a Record-A-Call conference can use the Hold, Transfer, Agent Help, or other features. However, if any inside party has enhanced speakerphone enabled, the enhanced mode will be disabled when the conference begins and must be re-enabled if wanted.

Record-A-Call is not supported on IP or SIP devices on P2P calls. For details, see “Peer-to-Peer Audio for IP and SIP Endpoints” on page 85.

### To use Record-A-Call with an Inter-Tel endpoint:

1. Press the Special button and dial the Record-A-Call feature code (385) or press the RECORD-A-CALL menu button. The display shows REQUESTING RECORD-A-CALL.

2. If required or allowed, enter the mailbox number. The display shows ENTER MAILBOX NUMBER. (If you do not enter a mailbox number, and you have a default Record-A-Call mailbox, the system automatically selects your default mailbox. If you do not have a default mailbox, the feature times out or can be canceled by pressing or ACCEPT.)

3. You hear a confirmation tone when the Record-A-Call feature is activated. The display shows RECORD-A-CALL IN PROGRESS.

4. To turn off Record-A-Call, repeat step 1 or end the call. The display shows RECORD-A-CALL TERMINATED.

### To use Record-A-Call with a single line endpoint:

1. During a call, hookflash and dial the Record-A-Call feature code (385).

2. If required or allowed, enter the mailbox number. The display shows ENTER MAILBOX NUMBER. (If you do not enter a mailbox number, and you have a default Record-A-Call mailbox, the system automatically selects your default mailbox. If you do not have a default mailbox, the feature times out or can be canceled by pressing or ACCEPT.)

3. You hear a confirmation tone when the Record-A-Call feature is activated. The display shows RECORD-A-CALL IN PROGRESS.

4. To turn off Record-A-Call, repeat step 1 or end the call. The display shows RECORD-A-CALL TERMINATED.
Agent Help

**IMPORTANT** The Agent Help premium feature is required to use Agent Help.

The Agent Help feature allows a user to request help from a designated “Agent Help Extension” during a two- or three-party call. When the request-for-help call rings, the Agent Help Extension can choose to join the call or reject the request. The Agent Help Extension can be a supervisor or other endpoint, an extension list, or a hunt group.

In DB Programming, an endpoint has two entries that can affect how this feature operates.

- The Agent Help Extension field determines the endpoint that is called when the Agent Help feature code is entered at the endpoint. This field can be set to any valid extension number for an endpoint, extension list, or hunt group, or it can be set to “None.”
- The User-Keyed Extension flag, if enabled, allows the user to select the endpoint that will receive the Agent Help request, even if a default Agent Help Extension is programmed as described in the item above.

**Table 24. Agent Help Extension and User-Keyed Extension Fields**

<table>
<thead>
<tr>
<th>Agent Help Extension</th>
<th>User-Keyed Extension</th>
<th>Result</th>
</tr>
</thead>
<tbody>
<tr>
<td>Ext. number</td>
<td>Disabled</td>
<td>The Agent Help Extension is called when the feature code is entered.</td>
</tr>
<tr>
<td>Ext. number</td>
<td>Enabled</td>
<td>After the feature code is entered, the user is prompted to enter the number wanted. If a number is not entered before the dialing timer expires, the programmed Agent Help Extension number is dialed automatically.</td>
</tr>
<tr>
<td>None</td>
<td>Enabled</td>
<td>After the feature code is entered, the user is prompted to enter the number.</td>
</tr>
<tr>
<td>None</td>
<td>Disabled</td>
<td>The user hears reorder tone if the feature code is entered.</td>
</tr>
</tbody>
</table>

When a user enters the Agent Help feature code, a private call is placed to the endpoint Agent Help Extension. While the call is ringing at the supervisor’s extension, neither the requesting user nor any other parties on the call can hear the private call ringing. If the supervisor answers the call, the system creates a conference to include the supervisor in the requesting user’s original call.

If the Agent Help Extension is an Inter-Tel endpoint, the microphone is muted, and the supervisor cannot be heard unless he or she presses the MUTE button. If the Agent Help Extension is a single line endpoint, the supervisor can be heard as soon as the conference is established. In either case, the supervisor can hear all other parties on the call.

Agent Help is not supported on IP or SIP devices in P2P calls. For details, see “Peer-to-Peer Audio for IP and SIP Endpoints” on page 85.

In a network setting, the Agent Help Extension does not need to reside on the same node as the user requesting Agent Help.
The type of endpoint determines how the Agent Help Extension is alerted:

- **Display endpoints**: If the Agent Help Extension is a display endpoint, the display shows that the incoming private call is an Agent Help request and identifies the user-name of the requesting station. The Agent Help Extension can choose to reject the call by not answering it or by entering the Agent Help Reject feature code (376). The call appears under the button (or a Call button if there is no button).

- **Non-Display endpoints and single line endpoints**: The Agent Help Request private call rings just as any private intercom call would; there is no way to distinguish it.

If the Agent Help Extension is an extension list or hunt group, the private call circulates as usual until it is answered. If an Agent Help request is not answered before the Forward No Answer timer expires, the request is considered rejected.

To signal to the other parties that the Agent Help Extension has joined the call, a system-wide Agent Help Tone flag can be enabled in DB Programming. If the flag is disabled, there will be no alerting tone. A timer, called the Agent Help Tone Interval timer, determines how often this tone is generated. If the timer is set to 0, the tone is generated only when the Agent Help Extension enters the call, a party is added to the call, or the call is placed on hold and retrieved.

If the Agent Help feature code is assigned to an Inter-Tel endpoint feature button with a lamp, the lamp status shows the following:

- The lamp flashes when the Agent Help Extension is being called.
- The lamp goes off if the Agent Help request is rejected or the feature is terminated.
- The lamp is lit solidly when the Agent Help Extension is in the conference.

When the requesting user hangs up, all parties are automatically disconnected. The Agent Help Extension can leave the call at any time, without affecting the other parties, by hanging up. If the other party (or parties) hangs up first, the requesting station and the Agent Help Extension remain connected in an intercom call. The requesting station can cancel the request (or remove the Agent Help Extension from the call) by re-entering the Agent Help feature code, thereby terminating the feature.

If necessary, the requesting endpoint in an Agent Help call can use the Hold, Transfer, Record-A-Call, or other features during the call, while the Agent Help Extension is connected. However, if any inside party has enhanced speakerphones enabled, the enhanced mode is disabled when the Agent Help conference begins and must be re-enabled if still appropriate.

The Agent Help feature will not function in the following cases:

- If the feature is not enabled in Premium Features, any user attempting to enter the Agent Help feature code will hear reorder tones and, if at a display endpoint, see a RESERVED FEATURE display.

- The Agent Help Extension cannot be in Do-Not-Disturb mode or have call forwarding enabled. If so, the requesting user hears reorder tones after entering the Agent Help feature code or entering the wanted extension number. The Agent Help request will not follow the forward or any programmed system forwarding.

- If the user who enters the Agent Help feature code is on a four-party conference call, the system will send reorder tones to signal that the Agent Help Extension cannot be added. There can be a maximum of four parties in a conference; attempting to add the Agent Help Extension would exceed the maximum.

- If there are no conference circuits available when the user enters the Agent Help feature code, the user will hear reorder tones and must try again later when circuits are available.
If the call is being monitored by a hunt group supervisor, the monitoring feature is terminated when the Agent Help Extension joins the call; an Agent Help conference call cannot be monitored.

- **To use the Agent Help feature with an Inter-Tel endpoint:**
  While on a call, press the Special button and then enter the Agent Help feature code (375). Users of Executive and Professional Display endpoints and Model 8560 endpoints can press the **AGENT HELP** menu button.
  
  a. *If you hear reorder tone*, the Agent Help feature is not available at your station, you already have four parties in your call, no conference circuit is currently available, or the Agent Help Extension is in Do-Not-Disturb.

  b. Depending on programming, you may be prompted to enter an extension number. If so, enter the extension number of the station you want to call for help. *If you do not enter a valid extension number*, you hear reorder tones and must start over. The display shows REQUESTING AGENT HELP.

  c. *If the Agent Help Extension accepts the call*, you will hear the Agent Help tone, if enabled. Display endpoints show **AGENT HELP IN PROGRESS**. When the Agent Help Extension hangs up, your display shows **AGENT HELP TERMINATED**.

  d. *If the Agent Help Extension rejects the call*, you will hear a confirmation tone. Display endpoints will show **AGENT HELP REJECTED**.

- **To use the Agent Help feature with a single line endpoint:**
  While on a call, hookflash and then enter the Agent Help feature code (375).
  
  a. *If you hear reorder tone*, the Agent Help feature is not available at your endpoint, you already have four parties in your call, no conference circuit is currently available, or the Agent Help Extension is in Do-Not-Disturb.

  b. Depending on programming, you may be prompted to enter an extension number. If so, enter the extension number of the endpoint you want to call for help. *If you do not enter a valid extension number*, you hear reorder tones and must start over. The display shows REQUESTING AGENT HELP.

  c. *If the Agent Help Extension accepts the call*, you will hear the Agent Help tone, if enabled. Display endpoints show **AGENT HELP IN PROGRESS**. When the Agent Help Extension hangs up, your display shows **AGENT HELP TERMINATED**.

  d. *If the Agent Help Extension rejects the call*, you will hear a confirmation tone. Display endpoints will show **AGENT HELP REJECTED**.

**To respond to an Agent Help request at a display endpoint:**
When you receive a private call from an Agent Help request, you will see a display that shows <name> REQUESTS HELP. You can do one of the following:

**To accept the call:** Answer as usual or press the **ANSWER** menu button. You hear the Agent Help Tone, if enabled. Your display shows **AGENT HELP IN PROGRESS**.

**To reject the call:** Enter the Agent Help Reject feature code (376) or press the **REJECT HELP** menu button. You hear confirmation tone.

**To place your endpoint in Do-Not-Disturb to block the call:** Enter the Do-Not-Disturb feature code (370) or press the **DND** button or the **IC CALL TO DND** menu button.
System Forwarding

System forwarding is a standard feature that provides the ability to program the system so that calls ringing at an endpoint will follow a database-programmed “forwarding path” that routes the call based on the type of call and the status of the intended endpoint. (The Call Forwarding endpoint feature can be used to override System Forwarding.) Several terms will be used to describe this feature, including:

- **Principal Station**: Original endpoint being called.
- **Forwarding Point**: A destination (endpoint, hunt group, or off-node device) to which the system-forwarded call is directed. Each forwarding path can contain up to four forwarding points.
- **Forwarding Path**: The sequence of forwarding points to which the call will be sent. Forwarding paths are programmed in the database. Up to 200 unique paths can be programmed. Each endpoint can have up to three forwarding paths.

Two system timers are used with this feature:

- **System Forwarding Initiate**: This timer determines how long a call will ring (unanswered) at the principal station before moving to the first forwarding point. The default value is 15 seconds, and the range is 2–255 seconds.
- **System Forwarding Advance**: This timer determines how long the call will ring unanswered before moving to the next forwarding point. The default value is 15 seconds, and the range is 2–255 seconds.

For each forwarding path assigned to the endpoint, the system checks the following three criteria to determine if and when a call should be forwarded:

- **Type of incoming call**: The following six different types of calls can be programmed to be sent to the forwarding path:
  - Outside calls received through a call routing table (including DID and E&M calls, but not including DISA calls)
  - Ringing outside calls
  - Transferred outside calls (including Automated Attendant and Voice Mail transfers)
  - Recalling outside calls
  - DISA calls (including DISA calls received through a call routing table)
  - Intercom calls
• **Endpoint status:** The system recognizes four different types of endpoint status.
  
  - *No Answer:* If the call is not answered at the principal station before the System Forwarding Initiate timer expires, the system sends the call to the forwarding path.
  
  - *Busy:* If the principal station is busy, the system immediately sends the call to the forwarding path. Both “No Answer” and “Busy” can be selected together to form a “Not Available” status.
  
  - *Do-Not-Disturb:* If the principal station is in Do-Not-Disturb, the system immediately sends the call to the forwarding path.
  
  - *Immediate:* The system immediately sends the call to the forwarding path. The principal station user cannot answer an immediately forwarded call, but can reverse transfer it from the forwarding point.

• **Day-mode/night-mode status:** The system day/night mode is used to determine if a forwarding path is taken. Either day mode, night mode, or both may be specified.

For an example, see the diagram below. In the example, the principal station user wants direct ring-in or transferred calls to ring at his or her endpoint during day mode. The principal station has a forwarding path with two forwarding points: a hunt group and Voice Mail. If the principal does not answer the call, it follows the forwarding path to forwarding point #1, a hunt group. If the hunt group does not answer the call, it would continue on the forwarding path and go to the principal station Voice Mailbox (forwarding point #2).

If the principal station is an Inter-Tel endpoint, the “ring principal once” option can be set that will signal the principal station when a call begins to follow the forwarding path. The signal to the principal station consists of a display (CALL SENT TO FORWARD PATH) and a single burst of ring tone. The call cannot be answered at the principal station, but can be reverse transferred from the system forward point.

If a call rings in to multiple endpoints, and one or more of those endpoints has system forwarding, the call will not follow any of the forwarding paths. Agent Help requests will not follow any forwarding path.

If a principal station or an endpoint forwarding point is a member of a hunt group, calls placed to the hunt group’s pilot number are unaffected by system forwarding. The hunt group calls will be received at the endpoint as usual and will not enter the system forwarding path.

A call follows only the forwarding path of the principal, even if a forwarding point has a forwarding path of its own. The call that originated at the principal station will follow only the principal station’s forwarding path.
Individual Endpoint Forwarding Points

The ringing duration at each endpoint forwarding point is determined by the System Forwarding Advance timer (defaults to 15 seconds). If the endpoint forwarding point is busy, the call will camp-on until the System Forwarding Advance timer expires and then it moves to the next forwarding point.

If the endpoint forwarding point is in Do-Not-Disturb, the forwarding path will bypass that endpoint and immediately send the call to the next forwarding point.

If a forwarding point is a voice processing system application, and the system is unable to communicate with the voice processing system, the call will bypass the forwarding point.

An endpoint forwarding point can place calls or transfer calls to the principal.

Hunt Group Forwarding Points

If a forwarding point is a hunt group, a call following the forwarding path will be sent to the announcement endpoint or overflow endpoint as usual; it will advance to the next forwarding point if not answered before the System Forwarding Advance timer expires.

The call will ring at the hunt group until the System Forwarding Advance timer expires. It will then move to the next forwarding point. The No Answer Advance timer determines how long the call will ring at each hunt group endpoint, as usual.

If all endpoints in a hunt group forwarding point have Do-Not-Disturb or hunt group remove enabled, the call will camp on until the System Forwarding Advance timer expires. The call will then be sent to the next forwarding point.

Handsfree Announce System Forward Option

When a call is transferred to the principal station by another endpoint user and the principal station’s system forward is set for the “no answer” endpoint status, a system-wide option determines what the caller hears, as shown in Table 25.

Table 25. Handsfree Announce and System Forward Options

<table>
<thead>
<tr>
<th>Handsfree Announce Option</th>
<th>System Forward Option</th>
<th>Result</th>
</tr>
</thead>
<tbody>
<tr>
<td>Disabled</td>
<td>Intercom and/or outside calls forwarded</td>
<td>The intercom call from the transferring endpoint enters the forwarding path after the System Forwarding Initiate timer expires. The transferring user hears ringing and can announce the call only if it is answered.</td>
</tr>
<tr>
<td>Enabled</td>
<td>Outside calls forwarded</td>
<td>The transferring user will place a handsfree call to the principal station and can announce the transfer; the System Forwarding Initiate timer does not start until the transfer is completed.</td>
</tr>
<tr>
<td>Enabled</td>
<td>Intercom calls forwarded</td>
<td>The handsfree announce option is overridden and the intercom call from the transferring station enters the forwarding path after the System Forwarding Initiate timer expires. The transferring user hears ringing and can announce the call only if it is answered.</td>
</tr>
<tr>
<td>Enabled or Disabled</td>
<td>Calls forwarded to Voice Mail</td>
<td>The transfer is completed when the transferring endpoint hangs up. The principal station's mailbox number is automatically dialed.</td>
</tr>
</tbody>
</table>
Manual Call Forwarding and System Forwarding

If the principal station or a forwarding point has manual call forward enabled (using the FWD button or one of the Call Forwarding feature codes), the Call Forward feature overrides System Forwarding.

If a forwarding point has the Call Forward feature enabled, a system-forwarded call will ring at the forward destination until it is answered or the System Forward Advance timer expires; then, the call moves on to the next forwarding point.
If a station forwarding point is manually forwarded to Voice Mail, the call will not follow the manual forward to the voice processing system; it will ring at the station forwarding point.

If the principal station receives a manually forwarded call (not a system forward), that call will not follow the principal station’s system forwarding path.

If an infinite forward loop results from the combination of manual forwards and system forwarding paths, the endpoint that was originally intended to receive the call will ring, even if the endpoint is in Do-Not-Disturb.
System Forward Enable/Disable

The user can enter feature codes that disable or enable System Forward at that endpoint. This flag does not affect the endpoint’s appearance on any other endpoint’s call forwarding path. The flag only determines whether the endpoint will have its own calls System Forwarded.

To enable or disable system forwarding:

- **Inter-Tel endpoints**: Enter the System Forward On (352), System Forward Off (353), or System Forward On/Off (354) feature code. The display shows SYSTEM FORWARD ON (or OFF). If disabled, the system ignores any System Forward paths programmed for this endpoint until System Forward is enabled.

- **Single line endpoints**: Lift the handset and enter the System Forward On (352), System Forward Off (353), or System Forward On/Off (354) feature code. If disabled, the system ignores any System Forward paths programmed for this endpoint until System Forward is enabled.
Unanswered System Forward Calls

If a call that is following a System Forward path is not answered or all forwarding points are in Do-Not-Disturb or busy, its final destination depends on the type of call, as shown in Table 26.

Table 26. Final Destinations for Types of Calls

<table>
<thead>
<tr>
<th>Type of Call</th>
<th>All FWD Points Busy or Do-Not-Answer</th>
<th>All FWD Points in Do-Not-Disturb</th>
</tr>
</thead>
<tbody>
<tr>
<td>Intercom</td>
<td>Rings at the last forwarding point until it is answered or the caller hangs up.</td>
<td>Call never leaves the principal endpoint.</td>
</tr>
<tr>
<td>Call routing ring-in (except DISA)</td>
<td>Rings at last forwarding point until it is answered or caller hangs up.</td>
<td>Rings at the principal endpoint until it is answered or the caller hangs up.</td>
</tr>
<tr>
<td>Direct ring-in</td>
<td>Rings at last forwarding point until it is answered or caller hangs up.</td>
<td>Rings at the principal endpoint until it is answered or the caller hangs up.</td>
</tr>
<tr>
<td>Transfer from an endpoint, Automated Attendant, or Voice Mail</td>
<td>Recalls the transferring endpoint's transfer recall destination when the System Forwarding Advance timer expires at the last forwarding point.</td>
<td>If immediate forwarding is enabled, rings the principal endpoint until the applicable Transfer timer expires. Then it recalls the transferring endpoint's transfer recall destination. If a conditional forwarding is enabled, rings principal endpoint until the System Forward Initiate timer expires. Then it recalls the transferring endpoint's transfer recall destination.</td>
</tr>
<tr>
<td>DISA (including Call Routing to DISA)</td>
<td>Rings at last forwarding point until the Abandoned Call timer expires.</td>
<td>Rings at the principal endpoint until the Abandoned Call timer expires.</td>
</tr>
<tr>
<td>Recalls</td>
<td>Recalls the principal endpoint's attendant when the System Forwarding Advance timer expires at the last forwarding point. If the principal endpoint does not have an attendant, the call continues to ring at the last forwarding point until the Abandoned Call timer expires.</td>
<td>If immediate forwarding is enabled, rings principal endpoint until the Recall timer expires. Then it recalls the principal endpoint's attendant. If a conditional forwarding is enabled, rings principal endpoint until the System Forward Initiate timer expires. Then it recalls the principal endpoint's attendant. If the principal endpoint does not have an attendant, the call continues to ring at the principal endpoint until the Abandoned Call timer expires.</td>
</tr>
</tbody>
</table>
Call Forward

With Call Forward, a user can route incoming intercom and outside calls (including direct ring-in calls) to another endpoint, to a hunt group, or to an outside phone number, if allowed by toll and trunk restrictions. In a network setting, the network allows the user to forward calls to endpoints or hunt groups on another node. (See also page 243.)

The four forwarding options are:

- **Forward all calls**: All incoming calls are immediately forwarded.
- **Forward if no answer**: Incoming calls are forwarded if they are not answered before the Forward No Answer timer expires.
- **Forward if busy**: Incoming calls are immediately forwarded if the endpoint is busy.
- **Forward if no answer or busy**: Incoming calls are forwarded immediately if the endpoint is busy or if calls are not answered before the Forward No Answer timer expires.

If forward all calls is enabled, display endpoints show the call forwarding status and destination until the request is canceled. If one of the conditional forwards is enabled (no answer, busy, or no answer/busy), display endpoints show the forward status for five seconds and then return to normal. If the endpoint receiving the forward is equipped with a display endpoint, it shows EXT XXXX FWD FROM EXT XXXX for each forwarded call received.

Users can chain Forwards from endpoint to endpoint provided that the Forwards do not form a logical loop and the chain does not include more than 10 endpoints. The conditional Forward features (if busy, if no answer, if busy/no answer) may form a loop that the system cannot detect until a call is placed to the forwarding endpoint. For example, if two users forward their calls to each other using the Forward If Busy feature, the system accepts the requests. However, if a call rings in while both endpoints are busy, the forwards create an illegal loop. In this case, the call camps on to the called endpoint and that endpoint’s display shows INVALID FORWARD PATH.

*If more than one endpoint has ring in for a trunk group*, direct ring-in calls on that trunk group will forward to extension numbers, but not outside numbers or Voice Mail ports. The display of the endpoint receiving the forwarded call will show it as a forwarded call, and the Call button or individual trunk button flashes to show ring in. The individual trunk button also flashes on the endpoint that is being forwarded.

Calls cannot be forwarded to restricted outside phone numbers or to endpoints in Do-Not-Disturb. *If the endpoint that is programmed to receive your forwarded calls is later placed in Do-Not-Disturb and you receive a call*, you will momentarily see a display showing that the destination endpoint is in Do-Not-Disturb; the call will remain at your endpoint. Direct ring-in calls that are forwarded to an endpoint in DND will ring the DND endpoint in accordance with the rules of DND.

*If your endpoint is in Do-Not-Disturb and you have call forwarding programmed*, the call is still forwarded, unless you enabled Forward No Answer. With Forward No Answer, intercom callers will see the programmed Do-Not-Disturb display, and the call will not be forwarded.
If calls are forwarded to a voice processing system application and the system is unable to communicate with the voice processing system PC, the call will not be forwarded. It will remain at your endpoint. Call Forward overrides System Call Forward at the principal endpoint.

If an endpoint is programmed for Forward No Answer or Forward No Answer/Busy and receives a transferred call, the Forward No Answer timer is started when the intercom call from the transferring endpoint begins ringing. The Forward No Answer timer is restarted if the transferring endpoint completes the transfer before the timer expires (the transferred call will then be forwarded when the timer expires).

Agent Help Request calls, queue callbacks, and recalls do not forward, except that a recall at an attendant’s endpoint will forward to another endpoint.

**FWD Button**

Some Inter-Tel endpoints use a combination of the default FWD key and forwarding menu keys. When the default FWD key is pressed, the display shows a menu of call forwarding options. The user must then select the option wanted by pressing the corresponding key next to the display. If the FWD key is programmed to use any other Forward feature code (355–358) the forwarding options will not be displayed when the FWD key is pressed; the Inter-Tel endpoint will operate the same as any other Inter-Tel endpoint.

Any Inter-Tel endpoint user can make any user-programmable button a FWD button that uses one of the forwarding feature codes (355–358). See the procedures on page 154 for programming user-programmable buttons.

A user-programmed FWD button is lit only when the endpoint is programmed for the call forwarding condition enabled by that button. For example, if a user-programmed FWD button is set to forward calls when the endpoint is busy, the button will be lit when the Forward If Busy feature is enabled, but not if the Forward If No Answer feature is enabled.

If an Inter-Tel endpoint user has both a fixed FWD button and a user-programmable FWD button, the fixed button will always light when the endpoint is forwarded. However, the user-programmable FWD button will be lit only when the forwarding option activated by that button has been selected. For example, if the user has the Forward All Calls feature programmed under a user-programmable button, that button will light if either the fixed or programmable FWD button is used to select that feature. However, if the fixed button or a feature code is used to set the Forward If Busy feature, only the fixed button will light.
Forward to an Outside Number

When programming an endpoint for call forward to an outside phone number, a trunk access code is programmed before the phone number. If the endpoint is called while the selected trunk group is busy, the call will not be forwarded. Calls will ring at the forwarded endpoint until the call is answered or the caller hangs up. The forwarded endpoint’s display shows \textit{FORWARD TO BUSY TRUNK}.

The forwarded endpoint’s (not the intercom caller’s) trunk and toll restrictions are checked when an intercom call is forwarded to an outside number.

When an outside call is forwarded to an outside number, the Unsupervised CO timer is activated. When the timer expires, the call recalls the attendant. If the attendant does not answer the recall before the Abandoned Call timer expires, the call is disconnected.

Endpoints with Forced Local Toll Call and Forced Long-Distance Toll Call account codes cannot forward calls to outside numbers.

If a trunk group is assigned direct ring in to multiple endpoints, and one of those endpoints is forwarded to an outside number, incoming calls on that trunk group are not forwarded to the outside number.

Depending on central office trunk quality, some reduction in voice volume may occur when an outside call is forwarded to an outside phone number.

Manual Forwarding to the Public Network

If disabled, the endpoint flag called \textit{Manual Forwarding to Public Network} prevents the endpoint from manually forwarding calls to the public network. This prevents users from calling in to a local endpoint and reaching toll numbers by way of the Call Forward feature.

Forward to the Message Center

Endpoint users can forward calls to their message centers. Inter-Tel endpoint users can forward calls to their designated message center by pressing the \textbf{FWD MSG} button or \textbf{MESSAGE CENTER} menu button. Single line users must use the Call Forward procedure and dial the extension number of the message center.

Forward to an Attendant

Endpoint users can forward calls to their Attendant by pressing the \textbf{FWD} button, or by entering a Call Forwarding feature code and dialing 0.

Forward to Voice Mail

If an endpoint is forwarded to Voice Mail or if the message center is a Voice Mail hunt group, the user’s mailbox number is automatically dialed when the Voice Mail unit answers the forwarded call.

If a chain of forwarded endpoints ends in Voice Mail, the mailbox number of the first endpoint in the chain will be selected when the Voice Mail unit answers the call.

If a trunk group is assigned direct ring in to multiple endpoints, and one of those endpoints is forwarded to a Voice Mail unit, incoming calls on that trunk group are not forwarded to the Voice Mail unit.
Forwarding Calls to an Extension or Outside Number

Endpoint users can also forward calls using the Remote Programming feature described on page 242.

To forward calls:

1. **Executive Display, Professional Display, and Model 8560 Endpoints**: Press the button. The display shows SELECT TYPE OF FORWARD. Press the menu button for the appropriate forwarding option (ALL, NO ANSWER, BUSY, or NO ANSWER/BUSY). The display shows ENTER FORWARD DEST.

2. **Inter-Tel endpoints**: While on- or off-hook, press the button or enter one of the following feature codes. (Display endpoints show ENTER FORWARD DEST.)
   - Call Forward All Calls (355)
   - Call Forward If No Answer (356)
   - Call Forward If Busy (357)
   - Call Forward If No Answer Or Busy (358)

**Single line endpoints**: Lift the handset and enter one of the feature codes listed above.

3. **If forwarding to an extension number or pilot number**, do one of the following:
   - Press a Mini-DSS or DSS/BLF button.
   - Dial the extension number or pilot number to receive the calls.
   - Dial 0 to forward to your attendant.
   - **(Inter-Tel endpoint users only)** Press the button or MESSAGE CENTER menu button to forward calls to your message center. **(Executive Display, Professional Display, and Model 8560 endpoint users only)** Press the button to forward to your mailbox. **(Executive Display, Professional Display, and Model 8560 endpoint users only)** Press the IC DIRECTORY menu button to look up and select a number.

If you dialed an invalid extension number, your display shows FORWARD TO INVALID NUMBER, you hear reorder tones, and must try again. If you dialed an endpoint that is in Do-Not-Disturb, the display shows DESTINATION IN DND, you hear reorder tones, and you must enter a new destination.
If forwarding to an outside phone number, do one of the following:

- Dial a trunk access code (8 or 92001–92208). The display shows ENTER PHONE #. Dial the phone number you want.

- (Inter-Tel endpoint users only) Press a trunk access button (OUTGOING, ARS, trunk group button, or trunk button). The display shows ENTER PHONE #. Dial the phone number. (Executive Display, Professional Display, and Model 8560 endpoint users only) Press the OUTSIDE PHONE menu button. The display shows ENTER PHONE #. Dial, speed dial, or redial the phone number or press the OUTSIDE DIR menu button to look up and select a number.

If the Forward destination is invalid because of toll restriction, ARS-Only restriction, no outgoing access for the trunk, no ARS or ARS route group for the dialed number, or an invalid dialed number, you hear reorder tones and must enter a new destination. The display explains why the call was denied.

4. Inter-Tel endpoints: If off-hook, hang up. You hear a progress tone, and the button is lit. The display shows FWD <condition> TO XXXX. If you attempted to forward calls to an endpoint that is unconditionally forwarded to your endpoint or if the forwarding path already has 10 endpoints, your display shows INVALID FORWARD PATH, you hear reorder tones, and you must enter a new destination.

Single line endpoints: Hang up.

To cancel any call forward request:

- Executive Display, Professional Display, and Model 8560 endpoints: Press the button and then press the menu button. You hear a progress tone and the button is unlit. The display shows ANY CALL FORWARD CANCELED.

- Inter-Tel endpoints: Press the lit button (the Speaker button lights) then press the Speaker button (both buttons go off). You hear a progress tone. Display endpoints show ANY CALL FORWARD CANCELED.

- Single line endpoints: Lift the handset, enter any Call Forwarding feature code (355–358), and hang up.
System Features

INTER-TEL® 5000 ADMINISTRATOR GUIDE – Issue 2.3, September 2007

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**Speed Dial**

The following three system features provide Speed Dial:

- **System Speed Dial**: Up to 1000 System Speed Dial numbers can be stored in the system. Each System Speed Dial number can contain up to 48 digits.

- **Station Speed Dial**: Each endpoint user can program up to ten 16-digit Station Speed Dial numbers. Display endpoint users can program a 16-character name for each Station Speed Dial number.

- **System Directory (Intercom and Outside)**: The intercom directory enables display endpoint users to “look up” intercom extension numbers and usernames. The outside directory enables display endpoint users to “look up” System Speed Dial numbers and associated names.

**System Speed Dial**

Speed Dial allows endpoint users to dial stored phone numbers quickly. Up to 1000 System Speed Dial numbers containing 48 digits each, can be stored with identifying names in system memory. See the System Directory feature on page 225 and Caller ID on page 137.

Each node in a network has its own System Speed Dial numbers. System Speed Dial numbers can be used only on the node where they are programmed.

To keep System Speed Dial numbers confidential, some or all can be programmed as non-display numbers. Non-display numbers can be used by any endpoint user, but are displayed only on the programming station’s endpoint. Nondisplay numbers cannot be redialed or saved as Station Speed Dial numbers at a display endpoint. Nondisplay numbers will appear in the SMDR record.

System Station Speed Dial names can be programmed by an administrator using English/Spanish or Japanese characters, or a combination. The programmed language for the endpoint does not affect the characters that can be viewed. That is, no matter what language the endpoint uses or which characters are in the name, the endpoint user will be able to see the Station Speed Dial names exactly the way they were programmed. In lists, such as Station Speed Dial directories, the English/Spanish characters are alphabetized before the Japanese characters. Japanese names will be at the end of the list.

System Speed Dial numbers are subject to toll restriction unless a system-wide option has been enabled that allows any endpoint to dial any System Speed Dial number regardless of toll restriction.

The System Speed Dial numbers are stored using location codes (000–999). When dialed, the numbers appear on a display endpoint unless they have been programmed as non-display numbers. Display endpoint users can also view System Speed Dial numbers and names without dialing; however, with non-display numbers, only the name is displayed.

System Speed Dial codes 000–999 can be stored in user-programmable endpoint buttons by following the directions given on page 154. These buttons allow one-button dialing of System Speed Dial numbers.

The System Speed Dial numbers and names are stored in battery-backed RAM and will not be erased in the event of a power failure.

The System Speed Dial numbers and names are programmed by the installer or at any administrator endpoint. For programming instructions, see page 50.
Viewing System Speed Dial Numbers

If you make a mistake, use the MUTE button to backspace.

To view System Speed Dial numbers and names:

1. While on-hook, press SYS SPD or enter the System Speed Dial feature code (381). The display shows REVIEW SYS SPEED # (000–999).
2. Enter the location code you want (000–999) or press the SEARCH menu button. The display shows the programmed name and number.
3. As appropriate, scroll forward through the Speed Dial locations in alphabetical order by pressing the high end of the Volume button or the NEXT button, or scroll backward by pressing the low end of the Volume button or the PREVIOUS button.
4. Repeat step 2 and step 3 for each Speed Dial location to be viewed. Or, press # or CANCEL to terminate this feature.
5. To automatically select a trunk and dial the displayed number, press #.

Dialing System Speed Dial Numbers

To dial System Speed Dial numbers at an Inter-Tel endpoint:

1. Lift the handset and select an outgoing trunk. (You may skip this step.)
2. Press SYS SPD or enter the System Speed Dial feature code (381). The display shows REVIEW SYS SPEED # (000–999). Non-display endpoint users hear a confirmation tone.
3. Enter the location code (000–999) for the number you want. If you have not selected a trunk, press # or the ACCEPT menu button to accept the number and automatically select a trunk. Or, press the ARS button, a trunk button, a trunk group button, or the OUTGOING button to select a trunk. The number is automatically dialed. Display endpoints show the number dialed unless it is a non-display number. If you hear reorder tones, your endpoint is restricted from dialing the selected number. If the display shows INVALID BIN NUMBER, you entered an invalid Speed Dial location code.

To dial System Speed Dial numbers at a single line endpoint:

1. Lift the handset and select an outgoing trunk.
2. Hookflash and enter the System Speed Dial feature code (381). You hear a confirmation tone.
3. Enter the location code (000–999) for the number. The number is automatically dialed. If you hear reorder tones, your endpoint is restricted from dialing the selected number.
Station Speed Dial

Each endpoint user can program up to ten 16-digit Station Speed Dial numbers. Single line endpoints use Speed Dial location codes (0–9). Inter-Tel endpoints use Speed Dial buttons, if programmed in the Inter-Tel endpoint keymap, or location codes 0–9. Together, the lamps in the Inter-Tel Station Speed Dial buttons create a busy lamp field that indicates the status of the endpoints programmed under the buttons, as summarized in Table 14 on page 81. Speed Dial buttons can contain outside phone numbers, feature codes, extension numbers, or hunt group pilot numbers.

Station Speed Dial codes can be stored in user-programmable buttons to create Speed Dial buttons.

Programming Station Speed Dial Numbers and Names

When entering the outside phone numbers, use one of the following methods:

- Use the dialpad to manually dial the number.
- On an Inter-Tel endpoint, press the **REDIAL** button to enter the last number (up to the first 16 digits) dialed or saved at the endpoint.
- On an Inter-Tel endpoint, press the **SYS SPD** button and enter the System Speed Dial location code (000–999) to store one of the System Speed Dial numbers in a Station Speed Dial location. If the number is over 16 digits, only the first 16 digits are stored. Non-display System Speed Dial numbers cannot be stored in Station Speed Dial locations.

If appropriate, outside phone numbers can be preceded with a trunk access code to allow one-button dialing of outside phone numbers. For example, a button programmed with “89619000” would select a trunk using the Outgoing Calls feature code (8) then dial 961-9000.

An endpoint extension number can be preceded with a pound (#) to always speed dial private intercom calls to the endpoint. Or, a “4” may be entered before an endpoint extension number or hunt group pilot number to reverse transfer (pick up) calls from that endpoint or hunt group.

If either of these options is used, normal handsfree intercom calls cannot be placed using that Station Speed Dial location or Speed Dial button and the Speed Dial button will not show the endpoint’s status.

An outside phone number can be preceded by a trunk access code for easier trunk selection and number dialing. Endpoint users can also program pauses and/or hookflashes [recalls in Europe] into the stored outside phone numbers. For example, the number can contain a specialized common carrier (SCC) local number, a pause, and an access code. When programming Speed Dial numbers, each hookflash and each pause is considered one digit. The durations of the hookflash and the pause are determined by the programmable CO Hookflash and Pause Dialing Digit Length timers.
Display endpoint users can program an associated name for each Station Speed Dial number. Speed Dial names can contain up to 16 characters. To program Speed Dial names, dialpad buttons are used to enter letters, numbers, and punctuation in English, Spanish and/or Japanese (Katakana). Only digital display endpoints can display Japanese characters. When programming Speed Dial names, you can press the \textbf{USE ALPHA MODE/USE NUMERIC MODE} menu button to switch back and forth between alphanumeric and numeric mode.

- **In numeric mode**, the dialpad buttons are used to enter numbers 0–9, the pound (\#) button is used for entering a hyphen (–), and the asterisk (*) button is used for entering a colon (:). For example, 1  00 would enter “1:00” in numeric mode.

- **In alphanumeric mode**, dialpad buttons are used to enter letters, numbers, and punctuation. The number of times a button is pressed determines which character is entered. For example, 533266 would enter “JEAN.” When adjoining characters are located under the same button, press the \textbf{FWD} button to advance to the next character. For example, 66  6667776 would enter “NORM.” To program information in alphanumeric mode using the letters printed on the dialpad buttons, see Table 27 on page 221.

**To program Station Speed Dial numbers at a display endpoint:**

If you make a mistake, press \textbf{CANCEL} and then start over. The number in memory remains unchanged. Or, use the \textbf{MUTE} button to backspace.

1. While on-hook, enter the Program Station Speed Dial feature code (383). The display shows \textbf{PROGRAM STN SPD (0–9) OR SCROLL}. Do one of the following:
   - **Use a Speed Dial button**: Press the Speed Dial button to be programmed.
   - **Use a location code**: Dial the applicable Speed Dial location code (0–9).
   - **Scroll through the displayed numbers**: Scroll to the number you want by pressing the Volume, NEXT, or PREVIOUS button.
   - **Use the Executive or Professional Display the Model 8560 endpoint menu**: Use the menu buttons to select a Speed Dial location.

2. When the display shows ENTER NAME do one of the following:
   - **To change or program the name**: Enter the name for the Speed Dial number using your dialpad as shown in Table 27 on page 221. Then press \textbf{ACCEPT}. \textbf{If you make a mistake}, use the \textbf{MUTE} button to back up and erase existing characters or the \textbf{CLEAR} button to erase the whole entry.
   - **To erase the current name and leave it blank**: Press \textbf{CLEAR} or press \textbf{MUTE} repeatedly until the name is erased. Then press \textbf{ACCEPT}.
   - **To leave the name the same**: Press \textbf{ACCEPT}. 

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**Page 220**  
**Programming Station Speed Dial Numbers and Names**
3. When the display shows ENTER NUMBER do one of the following:

- **To enter or change a number:** Dial the extension number or phone number (up to 16 digits, including pauses and hookflashes [recalls]) to be stored. To include *, #, a hookflash, or a pause in the number, press the Special button once for an asterisk (*), twice for a pound (#), three times for a hookflash [recall], or four times for a pause. You can enter more than one *, #, or hookflash in a row by pressing the FWD button to move to the next character (not necessary after pauses). If you make a mistake, press MUTE to back up and erase, or press ✗ or CANCEL to leave the number unchanged; then start over at step 1 on page 220.

- **To erase an existing number and leave it blank:** Press MUTE until the number is erased.

4. Press ✐ to exit. Repeat these steps for each additional number to be stored.

**Table 27. Programming Station Speed Dial Numbers Using the Dialpad**

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<th>Button</th>
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<td>ENGLISH/SPANISH CHARACTERS</td>
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</table>

* The Japanese characters are available only if the Multilingual feature is enabled and Japanese is installed as the secondary language.
To program Station Speed Dial numbers at a non-display endpoint:

If you make a mistake, press \# or CANCEL and then start over. The number in memory remains unchanged. Or, use the MUTE button to backspace.

1. While on-hook, enter the Program Station Speed Dial feature code (383).

2. Press the Speed Dial button to be programmed or dial the Speed Dial location code (0–9). If you hear reorder tones, you entered an invalid speed dial location code; try again.

3. Do one of the following:
   - **To enter or change a number**: Dial the extension number or phone number (up to 16 digits, including pauses and hookflashes) to be stored. To include *, #, a hookflash, or a pause in the number, press the Special button once for an asterisk (*), twice for a pound (#), three times for a hookflash, or four times for a pause. You can enter more than one *, #, or hookflash in a row by pressing the button to move to the next character (not necessary after pauses). If you make a mistake, press MUTE to back up and erase, or press \# or CANCEL to leave the number unchanged; then start over at step 1, above.
   - **To erase an existing number and leave it blank**: press MUTE until the number is erased.

4. Press \# to exit. Repeat these steps for each additional number to be stored.

To program Station Speed Dial numbers at a single line endpoint:

If you make a mistake, press \# and then start over. The number in memory remains unchanged.

1. Lift the handset and enter the Program Station Speed Dial feature code (383). You hear a confirmation tone.

2. Dial the Speed Dial location code (0–9). If you hear reorder tones, you entered an invalid speed dial location code; try again.

3. Do one of the following:
   - **To enter or change a number**: Dial the extension number or phone number (up to 16 digits, including pauses and hookflashes) to be stored. To include *, #, a hookflash, or a pause in the number, hookflash once for an asterisk (*), twice for a pound (#), three times for a hookflash, or four times for a pause. If you make a mistake, press MUTE to leave the number unchanged; then start over at step 1, above.
   - **To erase an existing number and leave it blank**: Overwrite the existing number by entering only a hookflash as the new number.

4. Hang up to exit. Repeat these steps for each additional number to be stored.
Viewing Station Speed Dial Numbers

To view the contents of a Speed Dial button (on a display endpoint):

1. While on-hook, enter the Review Keys feature code (396). The display shows PRESS THE KEY TO REVIEW. Press the applicable Speed Dial button. The display shows the currently stored number and name.
2. After viewing the number, press the Speaker button, ‑ or † to exit, or wait for the display to return to date and time.

To view the stored numbers and names (on a display endpoint):

1. While on-hook, enter the Program Station Speed Dial feature code (383). The display shows PROGRAM STN SPD # (0–9) OR SCROLL. Do one of the following:
   - **Use a location code**: Dial the wanted Speed Dial location code (0–9). When the name is displayed, press # or ACCEPT to view the number.
   - **Scroll through the displayed numbers**: Scroll to the number you want by pressing the Volume, NEXT, or PREVIOUS button. The name and number are displayed for each location.
   - **Use the Executive or Professional Display or the Model 8560 endpoint menu**: Use the menu buttons to select the appropriate Speed Dial location. When the name is displayed, press # or ACCEPT to view the number.
2. After viewing the number, press the Speaker button, ‑ or † to exit, or wait for the display to return to date and time.
Dialing Station Speed Dial Numbers

To dial a Station Speed Dial number at a single line endpoint:
1. Lift the handset. (Inter-Tel endpoints can remain on-hook.)
2. If placing an outside call, select an outgoing trunk or use ARS. If placing an intercom call or if the outside number contains a trunk access code, skip this step.
3. Hookflash and enter the Station Speed Dial feature code (382). You hear a confirmation tone. Then dial the location code (0–9). The number is dialed.

To dial a Station Speed Dial number at a non-display endpoint:
1. Lift the handset or remain on-hook.
2. If placing an outside call, select an outgoing trunk or use ARS. If placing an intercom call or if the outside number contains a trunk access code, skip this step.
3. Inter-Tel endpoint with a Speed Dial button: Press the Speed Dial button of the number you want. The number is dialed.
   Inter-Tel endpoint without a Speed Dial button: Enter the Station Speed Dial feature code (382). You hear a confirmation tone. Then enter the location code (0–9). The number is dialed.

To dial a Station Speed Dial number at a display endpoint:
1. Lift the handset or remain on-hook.
2. If placing an outside call, select an outgoing trunk or use ARS. If placing an intercom call or if the outside number contains a trunk access code, skip this step.
3. Inter-Tel endpoint with a Speed Dial button: Press the Speed Dial button of the number you want. The number is dialed. Display endpoints show the number.
   Inter-Tel endpoint without a Speed Dial button: Enter the Station Speed Dial feature code (382). Then do one of the following to dial the number:
   - Use a location code: Dial the applicable Speed Dial location code (0–9).
   - Scroll through the displayed numbers: Scroll to the number you want by pressing the Volume, NEXT, or PREVIOUS button.
   - Use the Executive or Professional Display or the Model 8560 endpoint menu: Use the menu buttons to select the applicable Speed Dial location.
Intercom, Speed Dial, and Feature Code Directories

Single line endpoints do not support the Directory feature.

The Directory feature enables display endpoint users to “look up” intercom extension numbers and usernames, System Speed Dial numbers and names, or system feature codes. Once the wanted information has been displayed, the user may automatically dial the number, store the number in a Station Speed Dial location, select a different number/name, terminate the directory routine, or switch to the other directory. A directory search can be performed when a call is waiting on conference or transfer hold at the station.

In a network setting, a user can see intercom directory names and extension numbers for all devices that are programmed as “off-node devices” on the local node. Devices that are represented by wildcard off-node extensions will not show up in the intercom directory.

To use the directory, the user enters a letter, a string of letters, a valid extension number, or a valid feature code. If searching for a name, the full name need not be entered. The system will find the closest match and show the number and its associated name on the endpoint display. Or, the user can scroll alphabetically through the stored list of names. (It is not possible to scroll through the extension numbers or feature codes numerically; extensions and features scroll alphabetically.)

Directory names can include English/Spanish or Japanese characters, or a combination. In the directories, English/Spanish characters are alphabetized before the Japanese characters. Therefore, Japanese names will be at the end of the list.

When using Japanese as the Primary or Secondary language, you can only search for last names in IC and CO directories. You cannot search based on the first name.

Keypad buttons are used to enter the English and or Spanish or Japanese (Katakana) letters, numbers, and punctuation. See Table 28 on page 226. Only digital display endpoints can be used for viewing Japanese prompts. The station user can switch between numeric and alphanumeric modes. (Numeric mode = button unlit; alphanumeric mode = button lit.)

In alphanumeric mode, the number of times a button is pressed determines which character is entered. For example, 77776444844 would enter “SMITH.” When adjoining characters are located under the same button, press the button once to advance to the next character. For example, 5666 66337777 would enter “JONES.” The letters correspond to the letters printed on the dialpad buttons.

The Message Lamp on Model 8500, 8520, and 8560 endpoints occupies the upper-right corner of the endpoint. If programmed for the Message feature, the lamp lights in alphanumeric mode.

The intercom directory is automatically updated whenever usernames and/or extension number information is reprogrammed. The outside directory is updated whenever a System Speed Dial number or name is reprogrammed.

Only administrator endpoints can view administrator feature codes using the directory. If the Diagnostics Mode feature is enabled, the diagnostic feature codes can also be viewed.
Use the following table as a guide for your entries using the keypad buttons.

**Table 28. Using the Dialpad to Locate Extension Numbers in the System Directory**

<table>
<thead>
<tr>
<th>Number of Times Button is Pressed</th>
<th>ENGLISH/SPANISH CHARACTERS</th>
<th>KATAKANA CHARACTERS*</th>
</tr>
</thead>
<tbody>
<tr>
<td>Button</td>
<td>1 2 3 4 5 6 7 8 9 10 11</td>
<td></td>
</tr>
<tr>
<td>Enveloped</td>
<td>A I U E O a</td>
<td></td>
</tr>
<tr>
<td>1</td>
<td>- &amp; ( ) 1</td>
<td></td>
</tr>
<tr>
<td>2</td>
<td>A B C ' 2</td>
<td></td>
</tr>
<tr>
<td>3</td>
<td>D E F ! 3</td>
<td></td>
</tr>
<tr>
<td>4</td>
<td>G H I &quot; 4</td>
<td></td>
</tr>
<tr>
<td>5</td>
<td>J K L # 5</td>
<td></td>
</tr>
<tr>
<td>6</td>
<td>M N O Ñ 6</td>
<td></td>
</tr>
<tr>
<td>7</td>
<td>P Q R S 7</td>
<td></td>
</tr>
<tr>
<td>8</td>
<td>T U V ? 8</td>
<td></td>
</tr>
<tr>
<td>9</td>
<td>W X Y Z 9</td>
<td></td>
</tr>
<tr>
<td>0</td>
<td>@ : . 0</td>
<td></td>
</tr>
</tbody>
</table>

* The Japanese characters are available only if the Multilingual feature is enabled and Japanese is installed as the secondary language.

**To perform a directory search:**

1. While on- or off-hook, enter the Directory feature code (307). The **MSG** button lights (and the Speaker button lights if on-hook) and the display shows **DIR TYPE**: (IC=1 OUT=2 FEAT=3).

2. Press one of the following:
   - **For Intercom Directory**: Press 1 or the **IC DIRECTORY** menu button to select the intercom directory. The display shows **INTERCOM NAME**.
   - **For Outside Directory**: Press 2 or the **OUTSIDE DIR** menu button to select the outside number directory. The display shows **OUTSIDE NAME**.
   - **For Feature Directory**: Press 3 or the **FEATURE DIR** menu button to select the feature code directory. The display shows **FEATURE NAME**.
3. Enter alpha and/or numeric characters for the number, code, or user/feature name (up to 10 characters for intercom directory, 16 for outside or feature directory). If searching for a name, the whole name does not have to be entered. You may switch between modes by pressing the \textit{MSG} button or the \textit{USE NUMERIC MODE/USE ALPHA MODE} menu button. If necessary, press \# to cancel the search.

   a. \textit{In numeric mode (MSG button unlit)}: Press the dialpad buttons to enter a valid feature, extension, or Speed Dial location number. Press \textit{MUTE} to backspace, if necessary. Press the \textit{SEARCH} menu button, or press \textit{MSG} to change to alphanumeric mode and then press \# to begin the search.

   b. \textit{In alphanumeric mode (MSG button lit)}: Press the dialpad buttons to enter the appropriate characters. See Table 28 on page 226. Press \textit{FWD} once to advance or twice to leave a space. Press \textit{MUTE} to backspace. Press \# or the \textit{SEARCH} menu button to begin the search. The closest match will be displayed. Executive Display, Professional Display, and Model 8560 endpoint users can return to the initial entry by pressing the \textit{START OVER} menu button.

4. If necessary, press the high end of the Volume button or the \textit{NEXT} button to scroll forward alphabetically through the directory or press the low end of the Volume button or the \textit{PREVIOUS} button to scroll backward.

5. Do one of the following:
   - \textbf{To store the selected number in a Station Speed Dial location}: Press the applicable Speed Dial button. You hear confirmation tone when the number is accepted.
   - \textbf{To automatically dial a displayed extension number}: Press \#, the \textit{IC} button, or the \textit{ACCEPT} button. (If an intercom call is camped on and the \textit{IC} button is lit, pressing the \textit{IC} button answers the waiting call and terminates the Directory feature.)
   - \textbf{To dial a displayed outside number}: Press \#, the \textit{CALL} menu button, or a Call button. The system automatically dials the number using your programmed outgoing access code.
   - \textbf{To dial a displayed feature code}: Press \# or the \textit{ACCEPT} button. The system automatically dials the feature code.
   - \textbf{To search for another name or number}: Repeat the procedure starting at step 3, above.
Intelligent Directory Search

The Intelligent Directory (IDS) feature requires Inter-Tel 5000 v1.2 or later.

The IDS feature simplifies searching for entries in a directory by significantly reducing the number of keystrokes required to find a match. However, because the dialpad has fewer buttons than the number of letters in the alphabet, each button represents several letters. For example, dialpad button 5 represents “5, J, K, L, j, k, or lower case L.” To enter JONES, you press 56537. As you press buttons, several possible matches appear on the display endpoint screen. This data entry process is similar to the “text on nine keys (T9)” feature found on some cell phones.

With minimum keystrokes, users can retrieve the names and phone numbers of persons entered into the Inter-Tel 5000 Intercom (IC) and Outside Directories and identify the status of features on endpoints from the Features Directory.

Voice Mail Directory instructions have not changed. The IDS feature does not apply to the Voice Mail Directory. Continue to use existing user guide instructions for entering and retrieving names from the Voice Mail Directory.

A display endpoint or an Inter-Tel® Model 8602 IP softphone application is required for using the IDS feature. The Model 8602 IP softphone is a software application that enables Voice over Internet Protocol (VoIP) telephone calls from laptop and desktop computers.

Pressing Dialpad Buttons to Create Characters

IDS allows you to retrieve entries from Inter-Tel 5000 IC, Outside, and Feature Directories.

Depending on the type of endpoint being used, you can create alphabetical characters by pressing the numeric and special character buttons on the dialpad. Digital and most IP endpoints require one method, which is explained in the following text. The Inter-Tel Model 8602 IP softphone requires a different method, which is explained in the Help files embedded with the application.
Creating Characters on Hard IP Endpoints

Inter-Tel 5000 v1.2 and later support dialpad button characters. When you need to search for a number, name, or feature entry in a directory, users can press the buttons shown in Table 29 to enter characters or perform an action. The buttons apply to buttons pressed on the endpoint dialpad.

Table 29. Dialpad Buttons-to-Characters Translation Table for Hard Endpoints

<table>
<thead>
<tr>
<th>Dialpad Button</th>
<th>Represented Character</th>
<th>Comment</th>
</tr>
</thead>
<tbody>
<tr>
<td>0</td>
<td>0</td>
<td></td>
</tr>
<tr>
<td>1</td>
<td>1</td>
<td></td>
</tr>
<tr>
<td>2</td>
<td>2A 2B 2C 2a 2b 2c Ç ą å á ÇÁ Åá</td>
<td></td>
</tr>
<tr>
<td>3</td>
<td>3D 3E 3F 3d 3e 3f é ê ë è É</td>
<td></td>
</tr>
<tr>
<td>4</td>
<td>4G 4H 4i 4i 4ii 4ii</td>
<td></td>
</tr>
<tr>
<td>5</td>
<td>5J 5K 5L 5j 5k l</td>
<td></td>
</tr>
<tr>
<td>6</td>
<td>6M 6N 6O 6m 6n 6o ô ö ò Ö ó ñ Ñ</td>
<td></td>
</tr>
<tr>
<td>7</td>
<td>7Q 7P 7R 7s 7r 7s</td>
<td></td>
</tr>
<tr>
<td>8</td>
<td>8T 8U 8v 8u 8ü ü</td>
<td></td>
</tr>
<tr>
<td>9</td>
<td>9W 9X 9Y 9Z 9w 9x 9y ý</td>
<td></td>
</tr>
</tbody>
</table>

<table>
<thead>
<tr>
<th>Dialpad Button</th>
<th>Action Characters</th>
<th>Comment</th>
</tr>
</thead>
<tbody>
<tr>
<td>*</td>
<td>Cancel</td>
<td>Returns endpoint to state before entering IDS.</td>
</tr>
<tr>
<td>#</td>
<td>Dial or perform</td>
<td>Dials or performs the first entry.</td>
</tr>
<tr>
<td>MUTE</td>
<td>Backspace</td>
<td></td>
</tr>
<tr>
<td>Menu Buttons</td>
<td>Dial or perform the entry or scroll a page</td>
<td>The Menu Buttons are located on the right or left side of the six-line LCD. Two-line display endpoints do not have Menu Buttons.</td>
</tr>
<tr>
<td>Volume Down</td>
<td>Previous match</td>
<td>Ignored if there is no previous match.</td>
</tr>
</tbody>
</table>
Finding an Entry in the IC Directory

The following sections show examples of display endpoint screens. You will see similar examples when using the IDS feature to retrieve an entry from the IC Directory. Also discussed are the actions you need to take when using the feature.

IC Directory Examples

Two-line display endpoints show only the first two lines of the six lines shown in the following examples. To access the Directory feature on a two-line display endpoint, press the Special button (SPCL or the infinity symbol) then press 307.

If you have not entered any digits in the IC Directory, you will see the first lines of the entire directory. Empty matches are displayed first. Instead of displaying a blank line, IDS shows the extension in brackets for any match that is empty. The following example of a six-line display screen shows that the first three IC Directory extensions of the system are not assigned in the database.

Scrolling Through IDS Entries

To scroll through the entries, press the >> arrow menu button or the Volume Up button to scroll forward, or press the << arrow menu button or the Volume Down button to scroll backward. The arrow menu button scrolls a screen full of entries at once, while the Volume Up and Volume Down buttons scroll one entry at a time. The arrow menu buttons at the bottom of the six-line display appear only when more than three entries are available. Also, if no previous page or entry exists, the << arrow menu button is unavailable and the Volume Down button does not respond. Likewise, if no next page or match exists, the >> arrow menu button is unavailable and the Volume Up button does not respond.

Dialing an IC Number from the IDS List

To dial the name in the first line, whose extension appears in the second line, press the pound (\#) button or the menu button next to the name. Menu buttons are available only on six-line display endpoints.
Finding Listings in the Outside Directory

Upon accessing the Outside Directory, you see the first entries the entire directory. If the directory is empty, a message appears indicating that entries are not available, and you hear a fast busy signal. If appropriate, phone numbers can be programmed as “Private” in the database and will appear as a “PRIVATE NUMBER” in the display, as shown in the following example.

Finding Features in the Feature Directory

The following examples depict Feature Directory displays. Upon accessing the Feature Directory, you see the first entries of the entire directory. Like the IC and Outside Directories, a six-line display endpoint shows three possible matches at a time. The entries on the first and third lines refer to the same extension. The menu buttons, arrow menu buttons, the total number of matches, underscore character, and dialpad buttons in the Feature Directory function the same as in the IC and Outside Directories.

In the following example, the user entered `SYS(353)` to search one of the System Forward features. For a list of default feature codes, refer to your endpoint user guide.
**House Phone**

This feature provides users with the ability to place a predesignated intercom or outside call simply by lifting the handset (or pressing the Speaker button, if using an Inter-Tel endpoint) on a designated House Phone. In a network, the House Phone can be programmed to dial an off-node device. Several applications for this feature include:

- **Courtesy paging endpoint**: Visitors hear pages instructing them to pick up the house phone (such as the paging endpoints used in airport terminals). When they lift the handset, they are connected to a pre-programmed station user who can give them a message or connect them to a call.

- **Emergency endpoint**: The House Phone can be programmed to automatically dial the Emergency Call feature code. Using the House Phone for this purpose could save time in an emergency.

- **Service endpoint**: Customers can use the House Phone(s) to place orders or receive special services from the lobby. For example, the House Phone would automatically dial the extension number of a service representative (or hunt group number of the service department).

- **Intercom network**: House Phones could be placed in specific locations throughout a building (such as examination rooms in a doctor’s office) and could be programmed to call a specific endpoint or group of endpoints (such as nurses’ stations).

Any endpoint may be designated as a House Phone in DB Programming. However, this feature is usually assigned to single line endpoints. The primary attendant should not be designated as a House Phone.

The number dialed by the House Phone is determined by the station’s Speed Dial programming. The number programmed through the database or in Station Speed Dial location 0 is automatically dialed during day mode, and the number in location 1 is dialed during night mode. This number can be either an extension number or an outside phone number. If it is an outside number, it must be preceded with a trunk access code (and a pause if necessary).

Once the House Phone status has been programmed, the Speed Dial number can only be changed while on-hook (if it is an Inter-Tel endpoint) or through individual endpoint information (special purpose endpoint) programming because lifting the handset will cause the endpoint to dial the designated number.

Incoming calls take precedence over outgoing calls. If using a single line endpoint or an Inter-Tel endpoint that is programmed for automatic trunk access (see page 157), any ringing call is automatically answered when the handset is lifted or the Speaker button is pressed.

The “House Phone Mode” flag determines whether a single line House Phone returns dial tone or does not return dial tone after the called party disconnects. At the System level of programming, all House Phones may be set in either Normal mode or Restricted mode. At the individual Endpoint level of programming, a House Phone must be programmed to dial specific digits as soon as the handset is taken off-hook. The programmed digits may ring a specific endpoint or a Hunt Group that rings multiple endpoints.

Normal mode allows the user to enter a feature code or place a call after the automatically called number hangs up. Restricted mode prevents the user from performing any operation other than placing a House Phone call. The System default state is Normal.
The interaction of the House Phone Mode flag with the programmed Speed Dial number and system feature is shown in Table 30.

**Table 30. House Phone Mode Interactions**

<table>
<thead>
<tr>
<th>Situation</th>
<th>DAY/NOIGHT Number</th>
<th>Normal Mode</th>
<th>Restricted Mode</th>
</tr>
</thead>
<tbody>
<tr>
<td>House phone user lifts the handset</td>
<td>Complete exten-</td>
<td>Dials the extension number</td>
<td>Dials the extension number</td>
</tr>
<tr>
<td></td>
<td>sion</td>
<td></td>
<td></td>
</tr>
<tr>
<td>House phone user lifts the handset</td>
<td>Blank</td>
<td>User receives intercom</td>
<td>User receives reorder</td>
</tr>
<tr>
<td></td>
<td></td>
<td>dial tone</td>
<td>tone</td>
</tr>
<tr>
<td>House phone user lifts the handset</td>
<td>Incomplete</td>
<td>System dials the partial</td>
<td>System dials partial number, then times out after Long Interdigit timer expires,</td>
</tr>
<tr>
<td></td>
<td>extension number</td>
<td>number and waits for further digits</td>
<td>and sends reorder tones</td>
</tr>
<tr>
<td>House phone is connected to ARS, and the user performs a hookflash</td>
<td>Dials an outside number using ARS</td>
<td>System inserts a hook-flash in the number and registers end-of-dialing</td>
<td>System restarts ARS by clearing the number and returns outside dial tone to the user</td>
</tr>
<tr>
<td>House phone is connected to an outside call, and the user performs a hookflash</td>
<td>N/A</td>
<td>The call is placed on</td>
<td>The trunk dials a hook-flash, and toll restriction is restarted</td>
</tr>
<tr>
<td></td>
<td></td>
<td>consultation hold, and the user hears intercom dial tone</td>
<td></td>
</tr>
<tr>
<td>Party the house phone is connected to hangs up before the house phone user</td>
<td>N/A</td>
<td>User receives intercom</td>
<td>System redials the house phone digits after the SL Wait for Disconnect timer expires</td>
</tr>
<tr>
<td></td>
<td></td>
<td>dial tone and can use another feature code or dial a number</td>
<td></td>
</tr>
</tbody>
</table>

**Redial**

The Redial feature can store one phone number in Redial memory of the endpoint. If the endpoint user reaches a busy number, is disconnected, or receives no answer, the number can be readily redialed. To redial, the user lifts the handset and presses the **REDIAL** button or enters the Redial feature code (380). A trunk access code is automatically entered and the phone number is redialed.

If redialing is activated while still connected to an outside call, the connection is dropped and the trunk is reseized before the number is dialed.

Only one phone number can be stored in an endpoint’s Redial memory at a time. This number can be stored in either of the following ways, depending on endpoint programming:

- **Last Number Saved**: The number is manually stored in Redial memory by the station user. Dialing other numbers does not change the stored number. It only changes when a new number is stored.

- **Last Number Dialed**: The last number manually dialed or speed dialed is automatically stored. It changes every time the user dials a phone number. Last Number Dialed is the default state of the Redial feature.
The Last Number Saved and Last Number Dialed features work differently depending on the station status when it is used, as shown in Individual endpoint programming determines the mode of the Redial feature, either Last Number Dialed or Last Number Saved. The Redial feature code (380) performs the Redial function and is programmed under the REDIAL button.

Non-display System Speed Dial numbers cannot be redialed at a display endpoint.

Table 31. Redial Interactions

<table>
<thead>
<tr>
<th>Endpoint Status When User Activates Redial</th>
<th>Endpoint Programmed for Last Number Saved</th>
<th>Endpoint Programmed for Last Number Dialed</th>
</tr>
</thead>
<tbody>
<tr>
<td>Idle</td>
<td>Saves the last trunk access code and phone number that was dialed.</td>
<td>A trunk is selected using the same trunk access code as used to place last outside call, and the last phone number is dialed.</td>
</tr>
<tr>
<td>Intercom dial tone (on- or off-hook)</td>
<td>Saves the last trunk access code and phone number that was dialed.</td>
<td>A trunk is selected using the same trunk access code as used to place last outside call, and the last phone number is dialed.</td>
</tr>
<tr>
<td>On an intercom call</td>
<td>Releases the current call, seize a trunk using the saved trunk access code, and dials the saved phone number.</td>
<td>Releases the current call, seize a trunk using the access code used on the last outside call, and dials the last phone number dialed.</td>
</tr>
<tr>
<td>After selecting a trunk but before dialing</td>
<td>Redials the saved phone number on the currently selected trunk, instead of the saved trunk access.</td>
<td>Redials the last phone number dialed, using the currently selected trunk.</td>
</tr>
<tr>
<td>After selecting a trunk and dialing one or more digits</td>
<td>Releases the current call, seize a trunk using the saved trunk access code, and dials the saved phone number.</td>
<td>Releases the current call, reseizes the trunk (using the access code used on that call), and redials the digits that were dialed.</td>
</tr>
<tr>
<td>On an incoming outside call</td>
<td>Releases the current call, seize a trunk using the saved trunk access code, and dials the saved phone number.</td>
<td>Releases the current trunk, seize a trunk using the same trunk access code used on the last outgoing call, and dials the last number dialed.</td>
</tr>
</tbody>
</table>
Using the Last Number Saved Feature

Use the following procedures to save and redial numbers at stations that are programmed to use the Last Number Saved feature. The saved number is replaced each time you repeat this procedure.

**To save a number on a station programmed for last number saved:**

*Inter-Tel endpoints:* While the endpoint is idle or while listening to intercom dial tone, press the **REDIAL** button, or press the Special button and enter the Redial feature code (380). You hear a confirmation tone, and the display shows LAST OUTSIDE NUMBER SAVED.

*Single line endpoints:* Lift the handset and enter the Redial feature code (380). When you hear the confirmation tone, hang up.

**To redial a number (last number saved):**

*Inter-Tel endpoints:* While on a call or after selecting a trunk, press the **REDIAL** button, or press the Special button and enter the Redial feature code (380). See Table 31 on page 234 to determine which trunk is selected. The saved phone number is dialed.

*Single line endpoints:* While on a call or after selecting a trunk, hookflash and then enter the Redial feature code (380). See Table 31 on page 234 to determine which trunk is selected. The saved phone number is dialed.

Using the Last Number Dialed Feature

Use the following procedure to redial numbers at stations that are programmed to use the Last Number Dialed feature.

**To redial a number (last number dialed):**

*Inter-Tel endpoints:* While on- or off-hook, press the **REDIAL** button, or press the Special button and enter the Redial feature code (380). See Table 31 on page 234 to determine which trunk is selected and which digits are dialed.

*Single line endpoints:* Lift the handset or hookflash. Then enter the Redial feature code (380). See Table 31 on page 234 to determine which trunk is selected and which digits are dialed.
Redirect Call

The Redirect Call feature code is located under System/Devices and Feature Codes/Feature Codes. The Redirect Call Feature (feature code 331) allows Inter-Tel endpoint users to route ringing outside, intercom, and camped on calls to another station, hunt group, or outside number. Routing of the redirected call is still subject to trunk and toll restrictions. This feature provides these options in addition to the currently available options which allow the station user to redirect calls to Voice Mail or Do-Not-Disturb.

The Redirect Ringing Call feature does not require premium features. Some specific aspects of this feature are described below.

- If the user has an Executive Display, Professional Display, or Model 8560 endpoint, the display will show a call menu option which reads SEND TO DEST. The user may choose this option or enter feature code 331. The display then prompts the user to enter the destination number directly or by scrolling through the IC directory.
- For non-display endpoints, calls are redirected using the feature code 331. This code is entered, followed by the appropriate destination number.
- If the endpoint receiving the redirected call is a display endpoint, it will show EXT XXXX FWD FROM EXT XXXX just as it does for a manually forwarded call.
- Calls may not be redirected back to the originating station. This will result in a message on the display saying CALL FORWARDED TO SOURCE. For non-display endpoints, the call will simply ring back at the station.
- Calls may not be redirected to restricted outside numbers, to stations in Do-Not-Disturb mode, or to stations that are forwarded. Redirected calls will not follow call forwarding.
- Agent Help Request calls, queue callbacks, and recalls cannot be redirected (these types of calls do not follow call forwarding).
- Calls cannot be redirected to another station’s mailbox, because redirected calls will not follow forwarding.
- If the call is redirected to an unreachable destination, it will not be redirected and will ring back at the original station.
- If an ARS, trunk group, or trunk number is entered, the system will prompt the user to enter the destination phone number. Calls may not be redirected to a node number, but may be directed to a station on another node by entering the correct extension number.
- If the applicable Forwarding timer expires before the user completes the redirection process, the system terminates the call.
Paging

The Paging feature allows announcements to be made through endpoint internal speakers. Optional external paging equipment (amplifiers and paging speakers) may also be installed.

There can be up to 10 paging zones. When the system is in the default state, all Inter-Tel endpoints are assigned to paging zone 1 to provide an all-page zone. Inter-Tel endpoints, trunks, and the external paging port(s) can be assigned to any, all, or none of the paging zones, as appropriate.

Placing a large number of Inter-Tel endpoints in a paging zone may affect system performance. If system operation is affected when a page is placed to a particular page zone, remove some Inter-Tel endpoints from that zone or change to external paging for the area served by that page zone.

In a network, all Inter-Tel endpoints and trunks within a page zone must reside on the same node as the page zone. However, a page zone can contain external page ports on other nodes.

Because the external paging ports have intercom numbers (91000 and 91001), stations can place an intercom call to the external paging ports, instead of using the Page feature, to make a page over the external paging speakers.

Pages are not heard on Inter-Tel endpoints that have been removed from paging (using the feature code as described below), are in Do-Not-Disturb, are ringing, or are in use. Also, background music on endpoints and on external speakers connected to endpoint speaker leads is interrupted for pages. If a trunk in a paging zone is unplugged, the page cannot be completed.

If appropriate, Inter-Tel endpoint users can program feature buttons to select the paging zones 0–9 (or 0–49) using paging access codes 9600–9609 (or 9600–9649).

The Page timer limits the length of pages. If it is set to 0, pages are unlimited in length.

To make a page:

1. Lift the handset.
2. **Inter-Tel endpoints**: Press the PAGE button, or enter the Page feature code (7). The display shows SELECT PAGE ZONE # (0–9).
   **Single line endpoints**: Enter the Page feature code (7).
3. Enter the zone code (0–9 or 0–49) for the appropriate zone.
4. After the single progress tone, make your page before the Page timer expires. If you hear reorder tones, and the display shows PAGE ZONE NOT PROGRAMMED, there are no stations in the selected page zone. If the display shows PAGE ZONE BUSY TRY AGAIN LATER, the necessary system resources are busy or the paging zone is being used by another station. Users cannot camp on to or queue the paging system.
5. Hang up.
Remove from Paging

An Inter-Tel endpoint user can prevent the endpoint from receiving pages or allow it to receive pages again using these feature codes. If the endpoint is assigned to more than one page zone, all zones are removed or replaced; the user cannot switch between individual zones.

To halt or enable zone pages:

1. While on- or off-hook, enter the Page Receive On/Off feature code (325) to prevent the endpoint from receiving pages. You hear a confirmation tone and the display shows PAGE RECEIVE ON (or OFF).

2. If off-hook, hang up.

Do-Not-Disturb

Placing an endpoint in Do-Not-Disturb mode halts all pages, incoming intercom calls, camped-on calls, and transferred calls to that station. Queue callbacks, recalls, and direct ring-in calls are not blocked. Another user calling the station while it is in Do-Not-Disturb hears a repeating signal of four fast tones and a pause. Display endpoints show the Do-Not-Disturb message. The user cannot Camp On, but can queue or leave a message at the station.

Direct ring-in calls forwarded to a station in DND will ring according to the DND rules.

If appropriate, individual endpoints can be prevented from using Do-Not-Disturb by disabling the Do-Not-Disturb option in the database.

If a hunt group station is in Do-Not-Disturb, calls to the user’s hunt group do not cause the endpoint to ring, but the individual trunk button will flash if all other stations in the hunt group are busy, forwarded, have hunt group remove enabled, or are in Do-Not-Disturb. Hunt group announcement stations and overflow stations can use Do-Not-Disturb to block hunt group calls by using Do-Not-Disturb.

Inter-Tel endpoints may be given Do-Not-Disturb override permission. These stations, when reaching a endpoint in Do-Not-Disturb, can enter the Do-Not-Disturb Override feature code (373) to place a non-handsfree intercom call to the station. Single line endpoints cannot be enabled to use the Do-Not-Disturb Override feature.

When an endpoint is placed in DND, the user may select one of 20 system-stored messages that will appear on the top line of the display (unless Do-Not-Disturb is enabled while the user is on a call or off-hook, in which case message 1 is automatically selected). When an endpoint in Do-Not-Disturb is called by a display endpoint user, the caller sees the selected message.

Do-Not-Disturb messages are programmed in the database or using an administrator endpoint. Default Do-Not-Disturb messages are programmed as shown in Table 32.

Table 32. Do-Not-Disturb Messages

<p>| | | |</p>
<table>
<thead>
<tr>
<th></th>
<th></th>
<th></th>
</tr>
</thead>
<tbody>
<tr>
<td>01</td>
<td>DO-NOT-DISTURB</td>
<td>08</td>
</tr>
<tr>
<td>02</td>
<td>LEAVE A MESSAGE</td>
<td>09</td>
</tr>
<tr>
<td>03</td>
<td>IN MEETING UNTIL</td>
<td>10</td>
</tr>
<tr>
<td>04</td>
<td>IN MEETING</td>
<td>11</td>
</tr>
<tr>
<td>05</td>
<td>ON VACATION ’TIL [ON HOLIDAY ’TIL]</td>
<td>12</td>
</tr>
<tr>
<td>06</td>
<td>ON VACATION [ON HOLIDAY]</td>
<td>13</td>
</tr>
<tr>
<td>07</td>
<td>CALL ME AT</td>
<td>14</td>
</tr>
<tr>
<td></td>
<td></td>
<td>15</td>
</tr>
<tr>
<td></td>
<td></td>
<td>16</td>
</tr>
<tr>
<td></td>
<td></td>
<td>17</td>
</tr>
<tr>
<td></td>
<td></td>
<td>18</td>
</tr>
<tr>
<td></td>
<td></td>
<td>19</td>
</tr>
<tr>
<td></td>
<td></td>
<td>20</td>
</tr>
</tbody>
</table>
Do-Not-Disturb works as follows in a network setting:

- Each node has a list of Do-Not-Disturb messages that can be used only on that node.
- An Intercom caller sees other users Do-Not-Disturb status when calling across nodes.
- The network allows DND override across nodes.

The system has default Do-Not-Disturb messages in both the Primary and Secondary Language. However, messages that use Japanese characters can be reprogrammed only through a Model 8660 endpoint designated as an administrator endpoint. See “System Programming” on page 36.

The language of the messages seen by the station user, both when programming his or her station and when calling another station that is in Do-Not-Disturb, is determined by the station’s programmed language. That is, if a station is programmed for Japanese, the user will see only Japanese messages when programming a Do-Not-Disturb message. If that station calls an American English-programmed station that has selected Do-Not-Disturb message 02, the Japanese station’s user will see the Japanese version of message 02. Only digital display endpoints display Japanese prompts.

The second line of the message can be customized with a numeric/alphanumeric message of up to 16 characters (English/Spanish and/or Japanese characters). The customized message is entered as described below. When entering a customized Do-Not-Disturb message, you can press the button or USE ALPHA MODE/USE NUMERIC MODE menu button to switch back and forth between alphanumeric and numeric mode.

- **In numeric mode**, the dialpad buttons are used to enter numbers 0–9, the pound (\#) button is used for entering a hyphen (-), and the asterisk (*) button is used for entering a colon (:). For example, 100 would enter “1:00” in numeric mode. When programming a custom Do-Not-Disturb message, the station is automatically in numeric mode.

- **In alphanumeric mode**, dialpad buttons are used to enter letters, numbers, and punctuation. The number of times a button is pressed determines which character is entered. For example, 337744432999 would enter “FRIDAY.” When adjoining characters are located under the same button, press the button once to advance to the next character. For example, 666 would enter “MONDAY.”
Use Table 33 as a guide for your entries using the keypad buttons.

**Table 33. Programming in Alphanumeric Mode**

<table>
<thead>
<tr>
<th>NUMBER OF TIMES BUTTON IS Pressed</th>
<th>ENGLISH/SPANISH CHARACTERS</th>
<th>KATAKANA CHARACTERS*</th>
</tr>
</thead>
<tbody>
<tr>
<td>Button 1</td>
<td>2</td>
<td>3</td>
</tr>
<tr>
<td>Button</td>
<td>1</td>
<td>2</td>
</tr>
<tr>
<td>1</td>
<td>-</td>
<td>&amp;</td>
</tr>
<tr>
<td>2</td>
<td>A</td>
<td>B</td>
</tr>
<tr>
<td>3</td>
<td>D</td>
<td>E</td>
</tr>
<tr>
<td>4</td>
<td>G</td>
<td>H</td>
</tr>
<tr>
<td>5</td>
<td>J</td>
<td>K</td>
</tr>
<tr>
<td>6</td>
<td>M</td>
<td>N</td>
</tr>
<tr>
<td>7</td>
<td>P</td>
<td>Q</td>
</tr>
<tr>
<td>8</td>
<td>T</td>
<td>U</td>
</tr>
<tr>
<td>9</td>
<td>W</td>
<td>X</td>
</tr>
<tr>
<td>0</td>
<td>@</td>
<td>:</td>
</tr>
</tbody>
</table>

* The Japanese characters are available only if the Multilingual feature is enabled and Japanese is installed as the secondary language.

When using either mode, endpoint users may use the Speed Dial buttons and/or the **REDIAL** button to enter stored numbers or messages. Speed Dial numbers can be chained together when entering messages that require more than 16 keystrokes to create the message. When programming a message in Speed Dial memory, you can use the Special button to enter a hookflash (F) in place of the **TWO** button to advance or insert spaces, or to enter a pause (P) to change numeric/alphanumeric modes within the message. The Special button can also be used to enter * or # in the Speed Dial number. Users can also place their endpoints in Do-Not-Disturb using the Remote Programming feature described on page 242.

**To enable Do-Not-Disturb mode:**

1. **Inter-Tel endpoints:** While on-hook, press the **DND** button, or enter the Do-Not-Disturb feature code (370) or Do-Not-Disturb On/Off feature code (372). Display endpoints show SELECT DND MESSAGE # (01–20).

2. **Single line endpoints:** Lift the handset and enter the Do-Not-Disturb feature code (370) or the Do-Not-Disturb On/Off feature code (372). You hear a progress tone.
3. To select message 01 (DO-NOT-DISTURB):

   **Inter-Tel endpoints**: Press the Speaker button or lift and replace the handset. The DND button lights.

   **Single line endpoints**: Hang up.

   **To select any of the 20 preprogrammed messages**: Enter the applicable number (01–20), or press the SCROLL menu button to begin scrolling. Display endpoint users may scroll through the messages by pressing the high end of the Volume button or the NEXT button to go forward, or the low end of the Volume button or the PREVIOUS button to go backward (the display shows the selected message). If you enter an invalid message number, you will hear reorder tones, and the display shows INVALID DND MESSAGE NUMBER; try again.

4. **If appropriate, customize the second line of the message**: Use any combination of the following methods:

   - **Remain in numeric mode**: Press the dialpad buttons to enter the number. Use the pound button (\#) for a hyphen (-) and the asterisk button (*) for a colon (:). Inter-Tel endpoint users can press the FWD button once to leave a space, or press the MUTE button to backspace.

   - **Change to alphanumeric mode (Inter-Tel endpoints only)**: Press the USE ALPHA MODE button (the button lights) or the USE ALPHA MODE menu button, then enter the appropriate characters. See Table 33 on page 240. Inter-Tel endpoint users can press the FWD button once to advance or twice to leave a space, or press the MUTE button to backspace.

   - **Use Speed Dial and/or redial numbers (Inter-Tel endpoints only)**: In either numeric or alphanumeric mode, you can speed dial a number (using the Speed Dial button or feature code). In numeric mode you can press the FWD button to enter the stored characters. You may chain Speed Dial and/or redial numbers together.

5. Terminate programming:

   **Inter-Tel endpoints**: Press the Speaker or the ACCEPT button, or lift and replace the handset. The DND button lights. Display endpoints show selected message. Date and time appear on the second line if there is no customized message.

   **Single line endpoints**: Hang up.

   **To cancel Do-Not-Disturb**:

   **Model 8660 or 8662**: While on-hook, press the lit DND button. The button goes off, and the display shows DO-NOT-DISTURB OFF.

   **Other Inter-Tel endpoints**: While on-hook, press the lit DND button, or enter the Cancel Do-Not-Disturb feature code (371) or Do-Not-Disturb On/Off feature code (372). The button goes off, and the display shows DO-NOT-DISTURB OFF.

   **Single line endpoint**: Lift the handset, enter the Cancel Do-Not-Disturb feature code (371) or Do-Not-Disturb On/Off feature code (372), and hang up.
To place your busy endpoint in Do-Not-Disturb while a call is camped on:

If, while on a call, you hear Call Waiting signals and you want to let the caller know you do not want to be disturbed:

**Inter-Tel endpoints**: Press the **DND** button or press the Special button and then enter a Do-Not-Disturb feature code (370 or 372). The button lights. The display shows DO-NOT-DISTURB ON. The intercom caller is no longer camped on and receives Do-Not-Disturb indications.

**Single line endpoints**: This procedure will interrupt the call you are on. Hookflash/recall and enter the a Do-Not-Disturb feature code (370 or 372). The call you are on is placed on Consultation Hold. The intercom caller is taken off of Camp On and receives Do-Not-Disturb indications. To return to the call you interrupted, hookflash/recall twice.

---

**Do-Not-Disturb Override**

If your Inter-Tel endpoint is enabled for Do-Not-Disturb override and you hear Do-Not-Disturb tones when placing an intercom call, you may use the following procedure to break through Do-Not-Disturb and complete the call.

**Do-Not-Disturb override (Inter-Tel endpoints with this feature enabled only):**

When you hear Do-Not-Disturb tones while placing an intercom call, press the Special button and enter the Do-Not-Disturb Override feature code (373). If the called station is idle, the call rings as a private intercom call. If the called station is busy, your call camps on.

---

**Remote Programming**

The Remote Programming feature allows a user to place an endpoint in Do-Not-Disturb mode (DND) or forward the station’s calls, or change the password; either from another station or through DISA. Each station has a password to limit access to this feature. (See page 150 for information about DISA.)

In a network, Remote Programming can be performed on one node for any Inter-Tel endpoint that is programmed as an off-node device on that node. You must go through DISA for remote programming of any endpoints that are not programmed as off-node devices.

---

**Endpoint Password**

The endpoint password is used only for the Remote Programming feature.

**IMPORTANT**

To prevent unauthorized use of the call forward feature, all endpoints using Remote Programming should have a password. To make the passwords difficult to guess, they should not match the extension number or consist of one digit repeated several times.

The endpoint password can be up to 8 digits in length. The default password is the extension number of the endpoint. The password can be changed by entering the Program Station Password feature code at the endpoint or when using the Remote Programming feature. It can also be programmed through Individual Endpoint programming.

If the password is changed from an endpoint, the user will be prompted for the old password, then the new password, and will be asked to verify the new password. If it is programmed through Remote Programming, the user will already have entered a correct password. In this case, the user is prompted only for the new password and then asked to verify the password.
Should the verified password and new password not match, the old password will be retained and the programming session canceled.

**To change the station password from your station:**

1. Enter the Program Station Password feature code (392). You hear a confirmation tone. If you have a display, it shows ENTER PASSWORD.
2. Enter your current password, followed by #. (At default, it is your extension number.) If you have a display, it shows NEW PASSWORD.
   - If you enter an incorrect password, the display will show INVALID PASSWORD, and your endpoint will return to the idle state.
3. Enter the new password followed by ✪. You hear a confirmation tone.
4. Enter the new password again for verification followed by ✪. You hear a confirmation tone. If you have a display, it shows PASSWORD SAVED. Your station returns to idle.

**To change the endpoint password using remote programming:**

1. If calling from an outside endpoint, access the system through a DISA line. Enter the DISA security code (if applicable). You hear a confirmation tone.
   - If programming from another endpoint, lift the handset or press the Speaker button.
2. Enter the Remote Programming feature code (359). You hear a confirmation tone. If you have a display, it shows ENTER EXTENSION. Users of Executive and Professional display endpoints and Model 8560 endpoints can press the IC DIR button to use the intercom directory to look up a number.
3. Enter the extension to be programmed. You hear a confirmation tone. If you have a display, it shows ENTER PASSWORD.
   - If you entered an invalid extension number, you will hear reorder tones and can try again.
4. Enter the extension’s password followed by ✪. You hear a confirmation tone.
   - If you enter an incorrect password, the display will show INVALID PASSWORD. Outside callers will be returned to dial tone, and stations will return to the idle state.
5. Enter the Program Station Password feature code (392). You hear a confirmation tone. If you have a display, it shows NEW PASSWORD.
6. Enter the new password, followed by ✪. You hear a confirmation tone.
7. Enter the new password again for verification, followed by #. You hear a confirmation tone. If you have a display, it shows PASSWORD SAVED. Outside callers will hear dial tone, and endpoint return to the idle state.

**Remote Do-Not-Disturb Programming**

After each entry, non-display endpoint users will hear a confirmation tone. Outside callers using a DISA line will hear confirmation tone followed by DISA dial tone. Display endpoint users will not hear confirmation tones but will see a prompt asking for the next entry.

**To place a station in Do-Not-Disturb mode using the remote programming feature:**

1. If calling from an outside endpoint, access the system through a DISA line. Enter the DISA security code, if applicable. You hear a confirmation tone followed by DISA dial tone.
   - If programming from another station, lift the handset or press the Speaker button.
2. Enter the Remote Programming feature code (359). If you have a display, it shows ENTER EXTENSION. Model 8660 or 8662 endpoint users can press the IC DIR button to use the intercom directory to look up a number.
3. Enter the extension to be placed in DND. If you have a display, it shows ENTER PASSWORD. If you entered an invalid extension number, you will hear reorder tones and can try again.

4. Enter the extension’s password, followed by * Can. If you have a display, it shows ENTER FEATURE CODE.

If you enter an incorrect password, the display will show INVALID PASSWORD, and the call is disconnected. Station users will hear reorder tones.

5. To turn on Do-Not-Disturb:
   a. Enter the Do-Not-Disturb On feature code (370). (The On/Off feature code cannot be used in remote programming.) If you entered an invalid feature code, you will hear reorder tones and can try again.

   If the station you are programming is not allowed to use DND, you hear reorder tones, and the display shows CANNOT ACCESS RESERVED FEATURE. Outside callers are returned to dial tone, and stations return to the idle state.

   b. Enter the applicable Do-Not-Disturb message number (01–20). If you entered an invalid message number, you will hear reorder tones and can try again.

   c. Enter the optional second-line message text, if wanted, using any combination of the following methods:

      Remain in numeric mode: Press the dialpad buttons to enter the appropriate number. Use the pound button ( # ) for a hyphen ( - ) and the asterisk button ( * ) for a colon ( : ). Inter-Tel endpoint users can press the MUTE button once to leave a space, or press the button to backspace.

      Change to alphanumeric mode (Inter-Tel endpoints only): Press the MSG button (the button lights) or the USE ALPHA MODE menu button, then enter the appropriate characters. (Refer to the chart and instructions in your user guide.) Inter-Tel endpoint users can press the MUTE button once to advance or twice to leave a space, or press the button to backspace.

      Use Speed Dial and/or redial numbers (Inter-Tel endpoints only): In either numeric or alphanumeric mode, you can speed dial a number (using the Speed Dial button or feature code). In numeric mode you can press the button to enter the stored characters. You may chain Speed Dial and/or redial numbers together.

   d. Hang up to complete the programming.

6. To turn off Do-Not-Disturb: Enter the Do-Not-Disturb Off feature code (371). (The On/Off feature code cannot be used in remote programming.) The call is automatically disconnected when the feature code is accepted. If you entered an invalid feature code, you will hear reorder tones and can try again.
Remote Forward Programming

After each entry, non-display endpoint users will hear a confirmation tone. Outside callers using a DISA line will hear confirmation tone followed by DISA dial tone. Display endpoint users will not hear confirmation tones but will see a prompt asking for the next entry.

To forward a station using the remote programming feature:

1. **If calling from an outside endpoint**, access the system through a DISA line. Enter the DISA security code (if applicable). You hear a confirmation tone followed by DISA dial tone.
   
   **If programming from another station**, lift the handset or press the Speaker button.

2. Enter the Remote Programming feature code (359). If you have a display, it shows ENTER EXTENSION. Model 8660 or 8662 endpoint users can press the IC DIR button to use the intercom directory to look up a number.

3. Enter the extension to be forwarded. If you have a display, it shows ENTER PASS-WORD. *If you entered an invalid extension number*, you will hear reorder tones and can try again.

4. Enter the extension’s password, followed by ✅. If you have a display, it shows ENTER FEATURE CODE.
   
   If you enter an incorrect password, the display will show INVALID PASSWORD, and the call is disconnected. Station users will hear reorder tones.

5. **To turn on Call Forward:**
   a. Enter one of the following Call Forward feature codes. *If you entered an invalid feature code*, you will hear reorder tones and can try again.

   Call Forward All Calls 355
   Call Forward If Busy 357
   Call Forward If No Answer 356
   Call Forward If No Answer/Busy 358

   b. Enter the forwarding destination. This can be an extension number or a trunk access code followed by an outside phone number.

   *If the forward destination is an extension number*, programming is complete, and the call is disconnected automatically.

   *If the forward destination is an outside number*, hang up to complete programming.

   *If the station is not permitted to forward to the destination entered*, you will hear reorder tones and can try again.

   *If you hang up before entering a forward destination*, the programmed forward is canceled.

6. **To turn off Call Forward**: Enter 355 and then hang up instead of entering a destination. You hear a confirmation tone.
Default Endpoint

An endpoint user can enter one feature code that will return the following features to their default states. Returning an endpoint to default:

- Cancels the following features, if in effect:
  - Do-Not-Disturb
  - Manual Call Forwarding
  - Queue Request
  - Account Code for All Calls Following
  - Background Music
  - Ring Intercom Always
  - Headset Mode
- Restores the following features, if disabled:
  - Handsfree
  - Page Receive
  - Hunt Group Replace
  - System Forwarding
- Returns endpoint volumes to default levels
- Returns endpoints to the standard keymap
- If the endpoint is a member of one or more ACD hunt groups, logs into the hunt group(s)

This feature is especially useful for installers and troubleshooters who need to know exactly how a given station is programmed.

**To return the station to default:**

**Inter-Tel endpoints:** While on- or off-hook, enter the Default Station feature code (394). You hear a confirmation tone, and the display shows **STATION DEFAULTED.** If **off-hook,** hang up.

**Single line endpoints:** Lift the handset and enter the Default Station feature code (394). You hear a confirmation tone followed by dial tone. Hang up.

Hookflash [Recall]

A timed hookflash [recall in Europe] may be required for Inter-Tel endpoint and single line users to use certain telephone company or PBX features. The CO hookflash [recall] feature code sends a timed hookflash [recall] over the trunk when entered. A hookflash [recall] restarts the call cost display and toll restriction, plus it starts a new line in the SMDR printout. However, the call remains on the same Call button.

The Hookflash [Recall] feature can be enabled or disabled on a trunk group-by-trunk group basis. It can be used on any outside call, including conference calls.

The duration of the hookflash [recall] can be changed by reprogramming the LS/GS CO Hookflash timer (default value is 0.6 seconds). If the system is not installed behind a PBX, the Hookflash timer must be programmed to be compatible with the system’s trunks.

**To send a timed hookflash/recall while on a trunk:**

**Inter-Tel endpoints:** Press the Special button and enter Hookflash feature code 330.

**Single line endpoints:** Hookflash and then enter the Hookflash feature code 330.
Reminder Messages

This feature does not apply to single line endpoints.

Reminder messages are set, like an alarm clock, to signal an Inter-Tel endpoint at a specified time. The user can select the message and time up to 24 hours in advance. Twenty different reminder messages are available. These messages can be reprogrammed by the installer or using an administrator’s station. They are limited to 16 characters.

At the programmed time, the reminder message signals the station with eight short tones. A display endpoint shows the message until it is canceled; a non-display endpoint receives tones only. If the endpoint is busy, the user still hears the tones and the message displays for 10 seconds during the call, then the display returns after the user hangs up.

Reminder displays interrupt, but do not affect, programming.

<table>
<thead>
<tr>
<th>Reminder Messages</th>
<th>Default Reminder Messages</th>
</tr>
</thead>
<tbody>
<tr>
<td>01 MEETING</td>
<td>11 CALL ENGINEERING</td>
</tr>
<tr>
<td>02 STAFF MEETING</td>
<td>12 CALL MARKETING</td>
</tr>
<tr>
<td>03 SALES MEETING</td>
<td>13 CALL ACCOUNTING</td>
</tr>
<tr>
<td>04 CANCEL MEETING</td>
<td>14 CANCEL DND</td>
</tr>
<tr>
<td>05 APPOINTMENT</td>
<td>15 CANCEL CALL FWD</td>
</tr>
<tr>
<td>06 PLACE CALL</td>
<td>16 TAKE MEDICATION</td>
</tr>
<tr>
<td>07 CALL CLIENT</td>
<td>17 MAKE RESERVATION</td>
</tr>
<tr>
<td>08 CALL CUSTOMER</td>
<td>18 REVIEW SCHEDULE</td>
</tr>
<tr>
<td>09 CALL HOME</td>
<td>19 LUNCH</td>
</tr>
<tr>
<td>10 CALL CORPORATE</td>
<td>20 REMINDER</td>
</tr>
</tbody>
</table>

In a network, each node has a list of reminder messages that can be used only on that node.

The system has default reminder messages in both the Primary and Secondary Languages. However, messages using Japanese characters can be reprogrammed only through an administrator’s station. See “System Programming” on page 36.

The language of the messages seen by the station user is determined by the station’s programmed language. That is, if a station is programmed for Japanese, the user will see only Japanese messages when programming a reminder message.

Only digital display endpoints will display Japanese prompts. Other Inter-Tel endpoints do not support Japanese prompts.
To request reminder messages:

Lift and replace the handset to stop the process without selecting a message.

1. While on-hook, enter the Reminder Message feature code (305). Display endpoints show SELECT REMINDER MSG # (01–20).

2. **To select a specific message:** Enter the two-digit message code (01–20). (Display endpoints show the selected message and SCROLL OR ACCEPT.)

   **To scroll through the messages:** View the available messages by pressing the high end of the Volume button or the NEXT button to scroll forward or pressing the low end of the Volume button or the PREVIOUS button to scroll backward. Each message displays for 0.5 seconds before another can be selected. When the appropriate message is displayed, press  or ACCEPT.

3. Enter the time that you want to receive the message. (If using a station programmed for Japanese, these prompts will be reversed, and you will set the AM/PM before the hour and minutes.)
   a. Enter the hour and minutes using four digits (e.g., 0900 for 9:00).
   b. *If the system is set for 12-hour display format,* the display shows SELECT AM OR PM. Press  or the AM button for AM or press  or the PM button for PM. You hear a progress tone when it is accepted, and the display momentarily shows the selected reminder message and time.

   *If an invalid time is entered,* you hear reorder tones and must enter the time again. Display endpoints show INVALID TIME.

To receive a reminder message:

1. At the selected time, you hear eight short tones, and the message is displayed. Do one of the following:

   *If your station is idle,* go to the next step.

   *If you are on a call,* the message will display for 10 seconds. After you hang up, the reminder message returns and must be cleared.

   *If you are programming your Inter-Tel endpoint,* the programming function is interrupted. The message will display, but will not interrupt programming; you can continue. After programming is completed, the reminder message will return and must be cleared.

2. **To clear the message:** Remain on-hook and press  The display shows EXPIRED REMINDER MESSAGE CLEARED.

To cancel all of your reminder message requests:

While on-hook, enter the Cancel Reminder Message feature code (306). The display shows REMINDER MSGS CANCELED. You cannot view or cancel individual messages.
Record Keeping and Maintenance Features

The record keeping and maintenance features available include Call Cost Accounting and System Error/Message Printing.

Call Cost Accounting

For Estimating Purposes Only

The system's Call Cost Accounting feature is intended to provide a cost estimate that is applied to the various classes of calls. Due to the wide variation in charges among network carriers, the system's call cost calculation cannot be used as a prediction of actual charges. This feature can only be used as a management tool to estimate call cost.

The Call Cost Accounting feature estimates the cost of outgoing and incoming calls, displays it on the Inter-Tel endpoints, and prints it in the SMDR reports. The cost is based on the type of call, phone number dialed, the elapsed time of the call, the day of the week, and the time of day. A table in the database supplies the rates for all types of calls, including multiplicative factors for evening and weekend rate changes on outgoing calls. The equation for calculating call cost is:

\[
\text{Daytime Rate} \times \text{Multiplicative Factor} \times \text{Connect Time}
\]

Multiplicative Factor Adjustment

The multiplicative factor adjusts the daytime per-minute call cost for evening and weekend rates of outgoing calls. For example, the evening call cost multiplier is 0.65 if calls are 35% less expensive after 5:00 PM. The daytime rate (D), evening (E) multiplicative factor, and night/weekend (N/W) multiplicative factors are used on the following schedule.

<table>
<thead>
<tr>
<th>Time</th>
<th>Factor Example</th>
<th>Table 35. Multiplicative Factor Cost Adjustments</th>
</tr>
</thead>
<tbody>
<tr>
<td>8 AM TO 5 PM</td>
<td>N/W D D D D D</td>
<td></td>
</tr>
<tr>
<td>5 PM TO 11 PM</td>
<td>E E E E E E N/W</td>
<td></td>
</tr>
<tr>
<td>11 PM TO 8 AM</td>
<td>N/W N/W N/W N/W N/W N/W N/W</td>
<td></td>
</tr>
</tbody>
</table>

If call cost is set to zero, call cost will not display during the call and the SMDR printout will show $00.00.

In a network setting, the call cost shown on the display and SMDR output use the factors and rates for the node on which the trunk resides. In other words, if a station on Node 1 dials what is considered a long-distance number on Node 1, but the call is routed to Node 2 where the number is considered local, the Inter-Tel endpoint will use the local call cost rate from Node 2. The call rate used for calls between nodes will be based on the Network call rate.

All outgoing calls using a trunk that is not subject to toll restriction are classified as one of the following call types for call cost calculation (call cost type is programmed in the database):

- Free
- Local
- Toll Local
- Toll Long Distance
- Incoming
- Operator
- Network
- International
Station Message Detail Recording

Station Message Detail Recording (SMDR) is a system feature that provides a detailed record of outgoing and incoming calls. The system records only valid calls. Outgoing calls become valid when, depending on system programming, the Valid Call timer expires or polarity reversal is detected. Outgoing calls become valid immediately if placed on hold or transferred. Incoming calls are always valid immediately.

Endpoint call data can be recorded on a customer-provided printer or an alternate device, such as a magnetic tape or floppy disk. The voice processing system PC can also be used to capture Station Message Detail Recording (SMDR) information. The SMDR information is stored in a file on the voice processing system PC hard disk for later retrieval.

Selectable SMDR options include any combination of the following. (See page 249 for an definition of call cost types.)

- **Record All Incoming Calls**: Records all incoming calls that are answered, except DID/DNIS calls.
- **Record All Local Calls**: Records all calls that use the “local” call cost.
- **Record All Free Calls**: Records all calls that use the “free” call cost.
- **Record All Ring-in Diagnostics**: A ring-in message is recorded for every incoming call (whether answered or unanswered) to indicate how long it rang.
- **Record All Toll Local Calls**: Records all calls that use the “toll local” call cost.
- **Record All Toll Long Distance Calls**: Records all calls that use the “toll long distance” call cost.
- **Record All Operator Calls**: Records all calls that use the “operator” call cost.
- **Record All International Calls**: Records all calls that use the “international” call cost.
- **Record All DISA Calls**: Records all DISA calls.
- **Record All Conference Calls**: Records all conference calls. If all conference parties exit except one station and one outside party, the call is no longer considered a conference and, when completed, it will be recorded using the appropriate call cost for that outside call.
- **Record All DID/DNIS Calls**: Records all calls received through DID and DNIS.
- **Record All Trunk To Trunk Calls**: Records all calls made from one outside caller to another.
- **Record All Network Calls**: Records all calls placed to and received from network nodes.
- **Record All Off-Node Devices**: Records all calls placed to and received from off-node devices.
The service technician can determine whether absorbed digits, equal access digits, and/or toll field digits appear in the SMDR printout. As an example, assume the following number was dialed: 89 (other system’s trunk access code) - 10288 (equal access code) - 1 (toll field) - 602 (area code) - 961-9000 (seven-digit number). The following programming options can be used:

- **Suppress Absorbed Digits**: Absorbed digits (on local or PBX lines) will not appear in the report if this option is selected. In the sample above, the absorbed digits (the other system’s trunk access code) would be suppressed so that only 10288- 1-602-961-9000 appears.

  If absorbed digits are repeatable on a local line, the absorbed digits will not appear in the SMDR report even when repeated.

- **Suppress Equal Access Digits**: Equal access digits will not appear in the report if this option is selected. In the sample above, the equal access code field would be suppressed to print only 89-1-602-961- 9000.

- **Suppress Toll Digits**: When this option is selected, toll digits will not appear in the report. In the sample above, the toll field would be suppressed so that only 89-10288- 602-961-9000 prints.

Any combination of the above can be used. If all three fields are suppressed, only 602-961-9000 will print.

The service technician can also suppress or allow call information in the SMDR report “Dialed Digits” field. The following options are available:

- **Suppress Outside Party Number**: Caller information that is received through ANI or Caller ID will not appear if this option is selected.

- **Suppress Trunk Number**: Information received through DID or DNIS will not be included in the report if this option is selected.

The service technician can determine which equipped station(s) and/or trunks will be included in the report, and whether off-node devices will be included in reports. However, for incoming calls, DISA calls, conference calls, and/or ring-in diagnostics, all calls are recorded even when they involve stations not in the programmed station list.

SMDR can be programmed to record the elapsed time of calls in seconds (S=XXXXXX) or hours and minutes (HH:MM). If programmed to record elapsed time in seconds, the ELAPSED TIME field will show “S=XXXXXX” (XXXXXX represents the number of seconds) for calls up to 999999 seconds long. For calls lasting longer than 999999 seconds, ELAPSED TIME will show “HH:MM” (hours and minutes rounded up to the nearest minute).

The SMDR output report is printed as shown in Figure 14 on page 252. A page heading (with the day of the week, date, month, year and column headings) is generated just after midnight to show the change in date. A header is also printed after output from another system source (alarm, informative message, etc.), using the same output device, interrupts the SMDR output.
**Figure 14. SMDR Report Format**

<table>
<thead>
<tr>
<th>TYP</th>
<th>EXT#</th>
<th>TRUNK</th>
<th>DIALED DIGITS</th>
<th>START</th>
<th>ELAPSED</th>
<th>COST</th>
<th>ACCOUNT CODE</th>
</tr>
</thead>
<tbody>
<tr>
<td>XX</td>
<td>XXXXXXXXXX</td>
<td>&lt;28 characters&gt;</td>
<td>HH:MM HH:MM:SS</td>
<td>$XX.XX XXXXXXXXXXXX*</td>
<td></td>
<td></td>
<td></td>
</tr>
</tbody>
</table>

**TYP**

- Call type abbreviations for:
  - Free: 000
  - Local: LOC
  - Toll Local: TLC
  - Toll Long Distance: TLD
  - Operator and International: O/I*
  - DISA: DSA
  - Conference: CNF
  - DID/DNIS (non-DISA): DID
  - Incoming: IN
  - Network: NET

* If the Display "O/I" for Operator and International Calls flag is disabled, operator calls are displayed under "OP" and international calls are displayed under "INT."

**EXT**
The extension number (XXXXX) of the last endpoint to handle the call is shown. For a CO-to-CO call, this field shows the extension number of the second trunk involved. For a conference call, it shows the endpoint that initiated or answered the call. For an unanswered ring-in, it shows *****. An outgoing call that has been initiated by another trunk will show the initiating trunk’s number.

**TRUNK**
The extension number of the trunk used during the call is shown. For an IP network call (using the networking IPR Application), this field shows the extension number of the IP connection used for the call.

**DIALED DIGITS**

For an outgoing call: The first 28 digits of the telephone number are shown (if ARS was used to place the call, the modified number, not the dialed digits, are shown). A “>” at the end of the number indicates that more than 28 digits were dialed.

Some digits may be suppressed (for an explanation of suppressed digits, see page 251). For a conference call, this field shows the endpoint that brought the trunk into the conference unless the conference ends as a call with only one endpoint and one trunk or if all conference parties are put on individual hold (in which case the field shows the last party to handle the call).
For an incoming call: This field is determined by the service type of the trunk that was used for the call and whether the digits are being suppressed. If the information is not suppressed, it is included in the report as shown in Table 36.

Table 36. SMDR Report Format

<table>
<thead>
<tr>
<th>Service Type</th>
<th>Call Record Shows:</th>
<th>Ring-In Record* Shows:</th>
</tr>
</thead>
<tbody>
<tr>
<td>DID, DNIS, or <em>DNIS</em> (not DISA ring-in)</td>
<td>Trunk DID or DNIS Information</td>
<td>Ring... Trunk DID or DNIS Information</td>
</tr>
<tr>
<td>DID, DNIS, or <em>DNIS</em> with DISA ring-in</td>
<td>Blank if answered; DISA digits if unanswered</td>
<td>DISA calls do not generate a ring-in record</td>
</tr>
<tr>
<td>Caller ID, ANI, or <em>ANI</em></td>
<td>Outside Party ANI or Caller ID</td>
<td>Ring...Outside Party ANI or Caller ID</td>
</tr>
<tr>
<td><em>ANI</em>DNIS*</td>
<td>Outside Party ANI or Caller ID and Trunk DID or DNIS Information</td>
<td>Ring...Outside Party ANI or Caller ID and Trunk DID or DNIS Information</td>
</tr>
<tr>
<td>None</td>
<td>Blank</td>
<td>Blank</td>
</tr>
</tbody>
</table>

* Ring-in records appear only if ring-in diagnostic records are enabled as described on page 250.

START
This shows the time that the call became valid (see page 250 for an explanation). For an unanswered ring-in, it shows the time the call began ringing. It is shown in 24-hour time (00:00–23:59).

ELAPSED
This shows the call length from the START time (above) until disconnect or the length of time an unanswered call was ringing. If the option is enabled that shows call duration in seconds, calls up to 999999 seconds long appears as S=XXXXXX (XXXXXX represents the number of seconds) and calls lasting longer than 999999 seconds appear as HH:MM (hours and minutes).

COST
This is the approximate cost of the call ($XX.XX), based on the database information. If cost exceeds $99.99, it is printed without the decimal point ($XXXXX). If it exceeds $99999, $$$$$$ is shown instead. If there is no cost, the field is blank.

ACCOUNT CODE
This is a standard, forced, or optional account code (up to 16 digits). An optional account code overrides standard or forced account codes. The field is blank if no account code was used.

If an asterisk (*) appears at the end of the record, as shown in the sample, it indicates that the call was terminated because the system detected a disconnect signal from the trunk (that is, the caller hung up before the station user on the call).
System Error/Message Print

The system has a self-diagnostic feature that monitors the system. When a system fault or other noteworthy condition is detected, the software determines the impact of the condition and classifies it as a major or minor alarm or as other message type. If the Message Print option is enabled, the messages are sent to the designated output port.

The service technician may enable any combination of the error message types. The available message types include the following:

- **Information Messages**: These messages give information regarding system operation.
- **Warning Messages**: The warnings in these messages indicate that a condition exists which may affect system performance.
- **Alarm Messages**: Alarm messages indicate that a minor alarm has occurred, but that general system operation was not affected.
- **Severe Messages**: These messages indicate that a severe error occurred in the system.

The messages can be programmed to be sent to any output device attached to the system. Messages can be printed through only one system port at a time. The print width is adjustable (64, 80, or 132 characters) to accommodate different output devices. When a line of information is longer than the programmed width, the information is printed on subsequent lines and preceded with “+” to indicate the continuation.

The system database also has a reserved area for recording the history of messages to allow service personnel to request a report at any time. The message history area has enough memory to maintain a record of at least 50 of the most recently generated messages. When the reserved area is full, the more recent messages begin to replace the older ones so that the most recent messages are available.

In a network setting, each node has its own message print programming, message print output port, and message print output port backup. There should be a message print terminal at each node to monitor node and network performance and aid in troubleshooting.

- If a node’s Message Print output port is a node, the network sends Message Print records to the specified node.
- You cannot select a node as the message print output port backup.
- If Message Print output programming forms a loop, the system will send the output to the node’s backup Message Print port. For example, if the Message Print port on Node 1 routes to Node 2 and the Message Print port on Node 2 routes to Node 1, the configuration will cause an infinite loop. Message Print reports for Node 1 would be printed to the backup serial port on Node 2 and vice versa.
- A two digit node number is inserted in front of the sequence number to indicate the node where the message originated. The following is an example of the new message print string format for a message the originated on node 3:

  -03:001 - 15:30 03-20 M6009 WRN EG CP

- Corrupted Queue Was Cleared: HISTORY QUEUE
- Message Print will *not* route diagnostic messages that apply to another node, except during manual Message Print requests. Those messages will only be printed locally.
- Message Print will *not* route messages during a startup or reset. The system will wait one minute before starting to send messages. Startup messages will only be printed locally.
- Messages may not be printed in time sequence order on remote nodes. The order of an incoming message print depends on how the message print was routed to the destination node. Therefore, a message print could possibly arrive at the destination message print node out of order.
In addition to the messages, the following error indications will occur in the event of a system alarm:

- If a major system alarm occurs, MAJOR ALARM appears on all display endpoints. The warning may also appear on a single display endpoint if the endpoint is defective. If all endpoints on the system are inoperative, the alarm message appears on the administrator’s display endpoint and the Inter-Tel 5000 Base Server LCD panel.

- A minor alarm sends a message to the primary attendant and to administrators, if programmed to receive alarms.

Each message record includes the following fields:

- **Sequence Number**: This field shows the order in which the faults occurred. If there is a skip in a series of sequence numbers, it indicates that a message was logged that is not programmed to be printed.

- **Time**: The Time field shows the time that the fault occurred. This field shows the hours and minutes in 24-hour format as HH:MM.

- **Date**: The Date field shows the date that the indicated fault occurred. The date shows the month and date as MM-DD.

- **Fault Code**: The Code field holds the associated number of the indicated fault. This field is four characters in length and the first character is an “M” for messages or “A” for minor alarms. Each of the different system faults that the system is capable of detecting is assigned a unique number (M100–M999 or A001–A299) in the system database.

- **Type**: This field indicates the following:
  - *Alarms*: The field shows “ALARM” for system faults that have not stopped system operation.
  - *Information Messages*: The field shows “INF.” The condition requires no attention.
  - *Warning Messages*: The field shows “WRN.” The condition requires attention because it may affect system performance.
  - *Severe Messages*: The field shows “SVR.” The condition requires attention because it has affected or will affect system performance.

- **Realm**: This field *does not appear for Alarm messages*. This two-character field indicates the realm responsible for the message. The field will contain one of the following abbreviations:
  - *DB* indicates that the message applies to the system database. These problems can often be fixed in DB Programming.
  - *HW* indicates the messages concerns the system hardware. Typically these problems are corrected by adjusting or replacing the affected hardware.
  - *EG* indicates a software error which will, most likely, require the intervention or attention of an Inter-Tel software engineer.

- **Source**: This field *does not appear for Alarm messages*. This two-character field indicates the software source responsible for the message.
  - *CP* indicates that Call Processing is the source
  - *OS* refers to the Operating System
  - *SS* refers to the System Software
  - *AL* refers to a Voice Processing System link
• **Text:** The Text field holds a text message that identifies the particular fault indicated by the record. This field is at least 40 characters in length and may extend onto subsequent lines if necessary to present adequate information about the fault. The information printed on an additional line is preceded with the character “+” to indicate the continuation.

• **Auxiliary:** When necessary, additional information may be printed as part of a message on separate lines. This additional information is referred to as “auxiliary text.” To identify this text as pertaining to the previous message, each line is preceded with the “<“character.

### Network Time Protocol Support

By enabling the Network Time Protocol (NTP) support feature, the internal real-time clock (RTC) on the Inter-Tel 5000 platform is updated daily in accordance with an atomic clock that is identified in DB Programming.

With NTP functionality enabled, the Inter-Tel 5000 updates the RTC and/or Linux date and time when the following events occur:

- When powered up or rebooted, the system updates both the RTC and Linux date/time using NTP. Otherwise, the system updates the Linux date and time using the RTC.
- Daily at 12:15 AM, the system updates both the RTC and Linux date and time using NTP. Otherwise, the system updates the Linux date and time using the RTC.
- Whenever the NTP, Time Zone, Enable Daylight Savings Time configuration changes in DB Programming, the system updates both the RTC and Linux date and time using NTP. Otherwise, the system updates the Linux date and time using the RTC.
- Whenever the Time/Date configuration changes from DB Programming or administrator endpoint, the system updates both the RTC and Linux date and time with the new date and time. However, if NTP is enabled the new setting is overridden when the next successful NTP synchronization occurs.

### Power Fail Bypass Capability

The Inter-Tel 5000 Base Server contains relay contacts that can be used for power fail bypass. This feature allows one or two single line endpoints to connect to predetermined trunks in the event of a processor board failure or a power failure. The single line endpoints and CO trunks are required.

### Customized System Programming Reports

Customized system programming reports can be created through the database. The programmer can select the programming area to be reported and then select the information that will be included. When extension numbers are listed in a report, they can be sorted by name, extension number or circuit number. Reports can also be titled.
Audio Diagnostics

The Audio Diagnostics feature code (320) allows users to generate diagnostics information for audio problems. When the Audio Diagnostics feature is initiated, users are prompted to answer questions about the audio problems by pressing specific dialpad buttons. Based on those answers, the system generates Alarm 128 on the administrator endpoint. For details, see “Managing System and Network Alarms” on page 54.

The Audio Diagnostics feature can be accessed when the endpoint is idle or when the user is on an active call. The amount of diagnostics information that the endpoint provides to the system depends on the state of the endpoint when the feature is accessed. An active call produces more diagnostic information than an endpoint in an idle state. If users do not want to access the Audio Diagnostics feature while on a call, they can access the feature immediately after they hang up.

In the following scenarios, the diagnostics information is not captured:

- If the user accesses the Audio Diagnostics feature but does not respond to the prompts on the display, the feature times out after 30 seconds.
- If either the user or the other party terminates the call before completing the diagnostics, the feature is terminated.

Audio Problems

When the Audio Diagnostics feature code (320) is entered, the system prompts users to select a characteristic of the audio problem. Users with 6-line display endpoints can either press the menu button that corresponds to the displayed audio problem or enter the numeric codes for the audio problem on their dialpads. Users with 2-line-display endpoints, or no-display endpoints, must enter the numeric codes on their dialpads because they do not have menu buttons.

A list of the 12 audio problems and the numbers associated with them is provided in Table 37.

Table 37. Default Audio Diagnostics Audio Problem Numbers

<table>
<thead>
<tr>
<th>Audio Problem</th>
<th>Number</th>
<th>Audio Problem</th>
<th>Number</th>
</tr>
</thead>
<tbody>
<tr>
<td>Echo</td>
<td>01</td>
<td>Too Loud</td>
<td>07</td>
</tr>
<tr>
<td>Static</td>
<td>02</td>
<td>Distorted</td>
<td>08</td>
</tr>
<tr>
<td>One-Way Audio</td>
<td>03</td>
<td>Delayed Audio</td>
<td>09</td>
</tr>
<tr>
<td>No Audio</td>
<td>04</td>
<td>Clipping</td>
<td>10</td>
</tr>
<tr>
<td>Choppy Audio</td>
<td>05</td>
<td>Cross-Talk</td>
<td>11</td>
</tr>
<tr>
<td>Too Soft</td>
<td>06</td>
<td>Other</td>
<td>12</td>
</tr>
</tbody>
</table>

Although 12 predefined audio problems are available, only four can be programmed for the system. The default numbers of the four selected audio problems may also be changed.

Specifying Audio Direction

After selecting the audio problem, the system prompts the user to choose the direction of the audio problem. The following options are available:

- ONLY I HEAR IT (users with a 2-line display, or a no-display, endpoint press 1)
- ONLY THE OUTSIDE (users with a 2-line display, or a no-display, endpoint press 2)
- WE BOTH HEAR IT (users with a 2-line display, or a no-display, endpoint press 3)
Using Record-A-Call for Diagnostics

If the Record-A-Call feature has been programmed for the station, users can record the call while using the Audio Diagnostics feature (see page 258).

The Record-A-Call feature cannot be used on certain calls such as Agent Help, Station Monitor, Conferences, and Paging.

Collecting Audio Diagnostics Data

If your Inter-Tel service technician instructs you to collect this diagnostics data, perform a System History Freeze using the instructions on page 61.

Using the Audio Diagnostics Feature

Only endpoints with the Audio Diagnostics feature enabled can use the feature.

Only 6-line display endpoints have the menu buttons mentioned in these instructions. Users who have endpoints with a 2-line display, or no display, must use the numeric entries to use the Audio Diagnostics feature.

To use the Audio Diagnostics feature:

1. **If the endpoint is idle**, while on-hook, enter the Audio Diagnostics feature code (320).
   
   If you are on an active call, press ☐, then enter the Audio Diagnostics feature code (320).
   
   The display shows PLEASE SELECT AUDIO PROBLEM.

2. Do one of the following to select an audio problem:

   EITHER, Press the appropriate menu button.

   OR, Enter the two-digit numeric code (01–12) that corresponds to the audio problem (See page 257).

   The display shows PLEASE SELECT AUDIO DIRECTION.

3. Select the direction of the audio problem.

   EITHER, Press the appropriate menu button.

   OR, Press the dialpad button that corresponds to one of the following responses:

   - Press 1 for ONLY I HEAR IT.
   - Press 2 for ONLY OTHER PARTY.
   - Press 3 for WE BOTH HEAR IT.

   If the Record-A-Call feature is enabled for your endpoint, the display shows WOULD YOU LIKE TO RECORD CALL?
4. If the Record-A-Call feature is available, you have the option of recording the call.
   
   EITHER, Press the appropriate menu button.
   
   OR, Press the dialpad button that corresponds to one of the following responses:
   
   • Press 1 to record the call.
   • Press 2 to continue without recording.
   
   If the system cannot accurately record the call, the display shows CALL CANNOT BE COMPLETED.

5. Hang up to complete the Audio Diagnostics feature.

   When the Audio Diagnostics selections are completed, the system generates Alarm 128 and the administrator station displays ALARM #128 <extension number> AUDIO FRZ.

To learn which Audio Diagnostics codes have been enabled for the system, consult an Inter-Tel field technician.

**Responding to the Audio Diagnostics Feature**

To respond to Alarm 128:

Alarm 128 indicates that someone has completed the Audio Diagnostics feature. Contact your Inter-Tel service technician so that they can poll the information for troubleshooting purposes.

**Phantom Devices**

Phantom devices are virtual extension numbers that are created for users who are usually not physically in the office and who do not need endpoints to retrieve calls or messages. Phantom devices do not consume a physical hardware address and do not count against the system total-device count.

**Basic Characteristics**

Phantom devices:

• Cannot make or receive calls
• Cannot be programmed as system administrators
• Have a Voice Mailbox
• Have the ability to forward endpoint calls to a current location (such as to a cell endpoint, pager, or home endpoint)

If the phantom device has a Voice Mailbox, the extension appears in the Voice Mail directory and the system intercom (IC) directory. Voice mail indications and endpoint silent messages are stored in the phantom device information.

When created, phantom devices are automatically added to the All Endpoints extension list. Phantom devices can be forwarded and follow the rules for endpoint manual forwarding and system forwarding.

Phantoms in v2.2 and later DB Programming are defaulted to the not-in-DND status. When upgrading an old database to v2.2 or later, if there were phantom devices in the old database, their DND statuses will not be changed after the conversion.
Virtual Device Functionality

With v2.2 and later, phantom devices are fully functional virtual devices on the system. Phantoms can function with Unified Communicator (UC) to perform advanced call routing tasks without the need for a physical endpoint. They can also have a true status, such as idle, Do-Not-Disturb (DND), ringing, etc. so that they could be placed in hunt groups and actually ring. The statuses that do not apply to phantom devices include:

- Unplugged: Phantoms have no hardware and will always be online.
- Busy/Permanently Busy: Phantoms cannot answer a call.

To take full advantage of the enhanced phantom feature, the user must have Unified Communicator (UC) or another Open Architecture Interface (OAI) application such as Contact Center Suite or Attendant Console. Because phantom devices do not have corresponding hardware devices, all programming that cannot be done in DB Programming must be done from the OAI application.

Operational Limitations

An OAI application can have phantoms perform almost all endpoint operations. The exceptions are listed below. These exceptions must be enforced because phantoms do not have voice paths.

- Phantoms cannot answer calls from an OAI application.
- Phantoms cannot make calls from an OAI application.
- Phantoms cannot participate in a conference.
- Phantoms cannot reverse-transfer calls.

Some OAI applications, such as Attendant Console, can send feature code requests to devices. The following feature codes are not available for phantoms:

- Answer
- Auto Trunk Answer
- Barge-In
- Change Language
- Conference
- Emergency Call
- Group Listen On/Off
- Handsfree On/Off
- Headset Off
- Headset On
- Headset On/Off
- Hookflash
- Music
- Page
- Page Receive On/Off
- Ring Tone Selection
- Queue
- Record-A-Call
- Redial
- Reverse Transfer
- Seize Device
- Steal
- System Hold
- Transfer
- Transfer To Hold
System Capacities for Phantom Devices

Phantoms consume the same software resources as a regular endpoint or IP endpoint, with the exception of the voice paths. Therefore, they can have an impact on system performance. The default capacities for phantoms are shown in Table 38.

Table 38. Phantom Devices Default Capacities

<table>
<thead>
<tr>
<th>System Type</th>
<th>Maximum Phantoms</th>
</tr>
</thead>
<tbody>
<tr>
<td>CS-5200</td>
<td>75</td>
</tr>
<tr>
<td>CS-5400</td>
<td>175</td>
</tr>
<tr>
<td>CS-5600</td>
<td>250</td>
</tr>
</tbody>
</table>

Practical Uses

The following scenarios describe phantom configuration uses:

- A phantom can be configured as a stand-alone device. An employee could have a cell phone to use when they are not in the office but still need to be on the office communications system. The administrator could program a phantom for them and the employee (or administrator) could forward the phantom to the cell phone, using UC. That way, callers could still call an extension on the office switch and be able to get the employee on his or her cell phone. In the extreme case, all devices on the switch could be phantoms for users with cell or other phones that are not physically connected to the switch.

- A phantom can be configured as a monitoring device. For instance, a user may put a phantom device in a hunt group and program an endpoint button to map to that phantom extension. With this setup, a user can monitor when the phantom is ringing, etc., and even answer the call either via a reverse transfer or a secondary extension key.

- A phantom can be configured as a call-park extension. The user may want to set up one or two (or more) phantoms to be extensions at which a call can be parked.

- A phantom can be configured as a general mailbox because phantoms can have associated mailboxes. When a caller calls the phantom, if the phantom is set up to go to voice mail, the caller can leave a message in this “general” mailbox. However, this configuration is not much different from an unassociated mailbox, with the exception that a phantom mailbox can be accessed off-node, but an unassociated mailbox cannot. Note that unassociated mailboxes can be accessed off-node if a caller calls the appropriate voice mail application and enters the unassociated mailbox extension.
Voice Processing Features

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Introduction

This section describes the voice processing system features supported by Inter-Tel voice processing systems. Features are dependent on the voice processing system, the equipment used, and the site’s licensing. Contact your local Inter-Tel reseller to determine which voice processing system features are supported on your system.

Automated Attendant

The automated attendant is a programmable feature that can be used to provide an automated call answering service. Calls can transfer, forward, or directly ring in to an automated attendant.

For information on external voice processing systems, see “External Voice Processing Systems” on page 22.

Calls to Automated Attendant are processed as shown in the following illustration.

When an automated attendant answers a call, it plays a recording that gives dialing instructions. During or after the recording, the caller may then directly dial an endpoint extension number, Voice Mail access number (if there is no associated mailbox), or hunt group pilot number. Or, the caller may use the directory to look up the extension.
The caller then has the following options:

- **Dial an endpoint extension number**: If an extension number is dialed, the call is transferred to the selected endpoint. If ringback tones are enabled, the caller hears ringing while the call is being transferred. If ringback is not enabled, the caller hears music. If the called endpoint is forwarded, the call follows the programmed forward.

- **Dial a hunt group number**: When a hunt group number is dialed, the call is transferred to the selected hunt group. The call rings or circulates according to how the hunt group is programmed (linear or distributed). If ringback tones are enabled, the caller hears ringing while the call is being transferred.

- **Dial the Voice Mail application’s extension number**: The caller can reach the Voice Mail main greeting by dialing the application extension number (access number) assigned to the Voice Mail feature. The caller can then leave a message as a non-subscriber or access any of the Voice Mail subscriber features. (See page 288 for Voice Mail information.)

- **Use the directory**: If the extension or mailbox number of the party is not known, the caller can spell the name using the dialpad buttons and “look up” the number in the directory. (This option can be disabled in the database. Or, if there are no names recorded for the individual mailboxes or for the system’s extension IDs, this option will not be provided.) Directory names can be sorted by first or last name. For information about using the directory, see page 276.

- **Dial the operator access destination**: If the caller needs further assistance, dialing 0 will access the voice processing system’s programmed operator destination. Or, if the caller is on a rotary endpoint and cannot enter a digit, the call will be automatically transferred to the operator destination. (The operator access destination is programmed in the database. There can be separate destinations for day and night modes.)

**IMPORTANT**

Due to the natural characteristics of the trunk, the volume level of DTMF tones transmitted over the trunk may be substantially reduced before reaching the system and voice processing system. This natural degradation in tone volume may adversely affect the reliability of the Automated Attendant feature. Other factors which can affect Automated Attendant performance are trunk noise and the quality and strength of the DTMF tones generated by the off-premises endpoint.

**Automated Attendant Applications**

There are a number of different uses for this feature. Here are two examples:

- Direct ring-in calls to a busy Attendant could be forwarded to Automated Attendant (using the Call Forward If Busy feature or system forwarding).

- Calls could ring in directly to Automated Attendant application’s extension number when the system is in day and/or night mode.
Automated Attendant Recall Destination

When an endpoint receives a call that has been routed through Automated Attendant, the call is handled as a transferred call and the display shows TFR FROM <description>. If the call is not answered, the call recalls Automated Attendant’s recall destination. The recall destination is usually the Automated Attendant Recall Destination application which announces that the called endpoint is unavailable and allows the caller to choose to leave a message (if the endpoint has an associated mailbox) or dial another extension number.

CALL TO AUTOMATED ATTENDANT

Automated Attendant answers and transfers the call to the destination selected by the caller.

The party receiving the transfer does not answer before the Transfer Voice Processing timers expires.

Call goes to the Automated Attendant Recall Destination where caller selects option.

Call is sent to station’s associated mailbox

Call is sent to endpoint, hunt group, Voice Mail, or operator destination.

If the Recall Destination fails to answer a call, it is automatically sent to the recall destination’s programmed Attendant. If the call is not answered there, it is disconnected after the Abandoned Call timer expires.

If an invalid number is dialed, the caller is prompted to enter another number. If a caller fails to enter a number before the Inactivity Alarm timer expires, the caller is prompted again. The caller cannot access trunks or enter feature codes through the Automated Attendant application. Trunk access codes and feature codes are considered invalid numbers.

Automated Attendant Custom Audiotex Recordings

Custom audiotex recordings are made using the voice mail administrator’s mailbox as described on “Creating Audiotex Recordings” on page 71. Each recording is associated with a recording number and assigned to the application(s) in DB Programming or using the system administrator’s mailbox.
Call Screening Options

When a call is received by Automated Attendant or Call Routing Announcement and the caller enters an extension number, the programmed Transfer Method determines how the call will be transferred. The Transfer Method flags can be programmed in the database. If allowed in mailbox programming, they can also be programmed by the mailbox user. The available Transfer Methods include:

- **Announce Only**: The caller is asked to state his or her name. The call is transferred to the associated extension number. When the user answers the transfer, the voice processing system plays the caller’s name and completes the transfer.

- **Screened**: The caller is asked to state his or her name. The call is transferred to the associated extension number. When the called party answers the transfer, the voice processing system plays the caller’s name. The called party has the options of replaying the name, sending the call to Voice Mail if the extension has a mailbox, transferring the call to another extension, accepting the call, or rejecting the call.

- **Unannounced**: The call is transferred to the associated extension number without announcing the call. This method is applied in the default state.

In a network setting, the voice processing system can provide call screening for a destination extension on another node. However, the node where the voice processing system is connected must have an off-node device programmed for the destination extension and access to the remote node.

Depending on the Transfer Method programmed for your mailbox or extension ID, one of the following will occur when you receive a call transfer from Automated Attendant or Call Routing Announcement:

**To receive Announce Only calls:**

1. When the caller enters your extension number, the system asks the caller to record his or her name.
2. The voice processing system places a private call to your endpoint.
3. When you answer, the voice processing system plays “You have a call from <caller's name>,” and then it completes the transfer.

**To receive Screened calls:**

1. When the caller enters your extension number, the system prompts the caller to record their name.
2. The voice processing system places a private call to your endpoint.
3. When you answer, the voice processing system plays “You have a call from <caller's name>,” and then gives you the following options:
   
   a. **To accept the call:**
      
      Press 1 or the ACCEPT menu button. The call rings at your endpoint.

   b. **To replay the announcement:**
      
      Press 1 or the REPLY menu button.

   c. **To send the call to Voice Mail:**
      
      Press 2 or the SEND TO V-MAIL menu button. The caller is transferred to your mailbox and hears your personal greeting.
d. **To forward the call to another extension:**

   Press `*` or the FORWARD menu button, and then enter the extension number. The call is sent to the endpoint you designated. If that endpoint has screened or announced transfers, the caller’s name is played again for that endpoint.

e. **To refuse the call:**

   Press `#` or the REFUSE menu button, or simply hang up. The caller receives a recording that says you are not available and offers the option of leaving a message.

**To receive Unannounced calls:**

   This is the default method for transferring a call. The call is transferred to your endpoint without any announcement.

---

**Extension ID**

Extension IDs are used in conjunction with transferring calls through the Automated Attendant or using the Extension Directory. The extension ID allows callers to be transferred to endpoints and applications that do not have mailboxes. It also allows the endpoint or application to have a recorded name in the directory. This feature is set up and initialized with the Database (DB) Programming and Voice Mail applications.

If an Extension ID has been created in DB Programming for an endpoint extension number, either the principal owner of the extension or the Voice Mail Administrator must set up (initialize) the ID with a new password and record a name for use in the Extension Directory. At default, the extension ID password is the same as the extension number.

If the Extension ID has not been “initialized,” calls can still be transferred to the associated extension number. However, they cannot be accessed from the directory. If the Extension ID’s username has not been recorded, it cannot be heard when callers access the Extension Directory. The name must be recorded in order to fully initialize the Extension ID. See page 62 for more information on initializing Extension IDs.

---

**IMPORTANT**

To provide system security, all extension IDs should have a password. To make the passwords difficult to guess, they should not match the endpoint’s extension number or consist of one digit repeated several times. The default password should be changed the first time the user logs in.

---

Once a password has been set up and the name recorded, the extension owner may access Extension ID Options which allows the associated directory name and password to be modified.

If an Extension ID has not been created for an endpoint, callers using Automated Attendant cannot be transferred to that destination. Instead, these callers will receive a system recording notifying them of an invalid entry and are routed back to Automated Attendant’s main menu.

Individual extension IDs can be programmed as “Private” in DB Programming. A Private number is included in the directory and can be dialed if the caller knows the number, but the extension number information is not available to the caller; only the name is played.

Extension IDs can also be programmed as “Unlisted.” That means that the number will not be included in the directory, but can be dialed if the caller knows the extension number.

Calls transferred from the Automated Attendant or a Call Routing Announcement application to endpoints with extension IDs can be screened, announced, or unannounced. Programming flags determine the methods used for transferring calls to endpoints with Extension IDs. For details, see “Call Screening Options” on page 268.
Automatic Fax Detection

This feature is available only when using an external voice processing system, either EM or VPU.

With Automatic Fax Detection, Call Routing Announcement applications and mailboxes can be programmed to automatically route incoming fax calls to a specified extension or to an e-mail address.

If the feature is enabled, the mailbox or Call Routing Announcement Application will automatically listen for fax tone.

- A mailbox will listen for fax tones during the mailbox greeting and while a message is being recorded. If the mailbox detects fax tones, the call will be transferred to the specified extension or e-mail address. If the tones are detected after a recording has started, the call will be disconnected.

- Call Routing Applications will be able to detect fax tones during the greeting and up to time-out. The fax tone detection will also be disabled if the caller performs an action that removes them from the Call Routing Announcement (transferring to an extension, transferring to a mailbox, etc.).

If the fax card is busy when an incoming fax call is received, the call will be disconnected.

Fax E-mail Requirements

To send faxes to an e-mail address using the Fax-On-Demand feature, the Unified Messaging feature, described on page 294, must also be enabled.

Fax Card Requirements

To send and receive faxes, the voice processing system PC must have a fax card installed. This can be the same fax card that supports the Fax-On-Demand feature. The software will support up to eight fax ports.

Fax E-mail Message Format

The e-mail message sent by the Voice Mail has the following components:

- **To**: The TO line contains the recipient’s e-mail address.

- **From**: The FROM line contains the e-mail address of the voice processing system. If an E-mail Real Name has been programmed for the voice processing system, it will also be shown on this line.

- **Subject**: The SUBJECT line contains the line “Fax received from *source*.” The source will be the Call Routing Application number or the mailbox number from where the fax is received, for example: CRA 2505 or MB 1000.

- **Received On**: The RECEIVED ON line contains the date and time that fax was received.

- **FAX.TIF file attachment**: The file attachment appears as a file with the name FAX.TIF. To view the fax, the user must display the TIF file using graphics software.
Automatic Speech Recognition

The Enterprise® Messaging (EM) voice processing system supports Automatic Speech Recognition (ASR), which allows users to access voice mail, call routing applications, the automated attendant and the directories by issuing spoken commands instead of dialing digits on their endpoint. Because ASR is a very flexible feature, it can be enabled or disabled on several different levels. ASR also allows subscribers and non-subscribers to navigate automated attendant menus, call routing applications, and search the directories using voice commands. When ASR is enabled for a user’s mailbox, he or she can issue spoken commands to navigate the voice mail menus. ASR is currently supported on EM voice processing systems only.

If ASR is enabled on a voice processing system, users are prompted to speak or press a digit on the dialpad to complete a voice mail task. If ASR is not enabled, users are only prompted to press a digit. When ASR is not enabled, users can still navigate the menus and perform tasks by pressing the appropriate dialpad digits when prompted.

The following ASR guidelines are provided in the endpoint user guides. Instruct users to follow these guidelines for best results when using ASR.

- When completing a mailbox-related task, either say the command each time or press the dialpad digits each time. Pressing the dialpad digits will temporarily disable ASR for the current call.
- Use the handset when issuing spoken commands. Your endpoint’s speaker can pick up background noise and other sounds, which may cause the system to misinterpret a command.
- If your endpoint has a display, voice mail options are shown on the display. If you don’t hear the command word when the prompt is announced, look at the display for the appropriate voice command.
- When you use the dialpad digits to complete a task, you are sometimes prompted to press * to indicate to the system that you have finished entering digits. It is not necessary to say “Pound” when using ASR. The silence that follows your voice command replaces the * digit.
- While you are recording a message, ASR is temporarily disabled, therefore you cannot issue spoken commands to access recording options like pause and erase. If you attempt to issue a spoken command while you are recording, your recording will include the command you issued. To access recording options, you must press the appropriate dialpad digits.
- You cannot use ASR to set up your e-mail password. For security reasons, you must enter your e-mail password characters by pressing the appropriate dialpad digits. Entering passwords this way prevents others from overhearing your password.
- If the system cannot interpret your voice commands for three consecutive attempts, ASR is temporarily disabled by the system for the current call. You can still complete tasks using the dialpad digits; however, you will no longer be prompted to say a command.
- You can temporarily disable ASR for the current call by selecting option eight from the main menu.
Call Routing Announcement

The Call Routing Announcement application can be used as a simple playback device that plays a message and then hangs up to disconnect the call. This is especially useful for programming hunt group announcement and overflow endpoints. Or, the Call Routing Announcement application can use digit translation which allows the caller to press a single digit for access to an endpoint, hunt group, or mailbox.

When a Call Routing Announcement application is used as a hunt group announcement or overflow endpoint, calls to the application automatically stop circulating through the hunt group if the caller selects a valid digit translation option. This allows the application to send the call to other endpoints without the call being “pulled back” into the hunt group when a hunt group endpoint becomes available and answers the call. However, if the caller does not dial a valid digit translation option, the call will be pulled back if a hunt group member answers.

Call Screening

Calls transferred from the Automated Attendant or a Call Routing Announcement application can be screened, announced, or unannounced. Separate programming flags determine the methods used for transferring calls to endpoints with mailboxes, endpoints with extension IDs, and extensions without mailboxes or IDs. For details, see “Call Screening Options” on page 268.
Custom Audiotex Recordings

When accessed, the Call Routing Announcement application plays a custom audiotex recording and then disconnects.

Call routing announcements are recorded using the Voice Mail Administrator’s mailbox as described on “Creating Audiotex Recordings” on page 71. Each announcement is assigned to use a “custom greeting” number in DB Programming.

Recordings for fax documents should include all dialing instructions. The recording should state whether documents can be selected by number and/or list all options. If fax delivery times are set to specific days or times, the recording should also include this information.

Digit Translation

Digit translation allows callers to dial a single digit to access a designated extension number, mailbox, or hunt group. Up to 12 digit translation storage locations (#, *, 0–9) are available for each Call Routing Announcement application. To use digit translation, the Call Routing Announcement application must have special programming.

After the application is created and programmed, the Voice Mail Administrator’s mailbox is used to make a custom audiotex recording that explains the options to callers, as described on “Creating Audiotex Recordings” on page 71. The custom audiotex recording is then assigned to the Call Routing Announcement application. A caller who listens to the Call Routing Announcement recording can dial the single-digit location number to reach the designated destination. For example, if the hunt group extension for a customer support group is programmed for digit 3, the recording should say something like, “For customer support, press 3.”

The system provides a choice between Primary and Secondary Language prompts and displays. The available languages are American English, British English, Spanish, and Japanese. The system selects the language to use for each call, as determined by the trunk, endpoint, and voice processing system programming. When using a Call Routing Announcement application with digit translation, the individual voice processing system applications assigned to the digits can be programmed to override the device language and provide prompts in one language only. Using this method, you can program a Call Routing Announcement tree that offers callers a choice between languages. For more information, see “Language Selection” on page 159.

Digit Translation Nodes

A “digit translation node” is a digit translation destination that allows further digit translation options. Unlike extension and mailbox destinations, a digit translation node does not move the call to a specific location. Instead, it offers additional dialing options. A digit translation node can use any of the Actions, including transfers to other digit translation nodes.

The use of digit translation nodes allows the programmer to create a Call Routing Announcement with several layers to form a digit translation “tree.” The tree can be made up of digits with destinations and/or digit translation nodes with additional translations. Figure 15 on page 274 shows an example of a Call Routing Announcement application with a multi-layered tree.
**Figure 15. Sample Call Routing Announcement Tree**

EXT 2500: “Thank you for calling X Company, if you know the party’s extension number, you may dial it now. For a company directory, press #. For Sales information press 3. For technical support, press 4.”

**CALLER DIALS #**
- Company Directory

**CALLER DIALS A NUMBER THAT BEGINS WITH 1**
- Call is sent to extension number dialed

**CALLER DIALS A NUMBER THAT BEGINS WITH 2**
- Transfer to Sales Node: “To speak to the operator, press 0. To select Sales information by fax, press 1. To speak to a Sales representative, press 2.”

**CALLER DIALS 3**
- Transfer to Hunt Group 2001 (Tech Support)

**CALLER DIALS 4**
- Transfer to Node (Sales Info Fax): “If you know the fax document number, please dial it now. If you want to speak to the operator, dial 0. For a product list, dial 2. For a price list, dial 3. For training information, dial 4. Then press # to check your selections and enter your fax number or press * to cancel.”

**CALLER DIALS 0**
- Transfer to Operator

**CALLER DIALS 1**
- Transfer to Extension 2000 (Sales Hunt Group)

**CALLER DIALS 2**
- Fax with that number is selected.

**CALLER DIALS 3**
- Selects Fax Document 1999 - Product List

**CALLER DIALS 4**
- Selects Fax Document 1998 - Price List

**CALLER DIALS #**
- Selects Fax Document 1997 - Training

**CALLER DIALS ***
- End Fax Selections: System requests dialing information and send fax, then returns to Main Menu.
- Cancel Fax Selections: System returns to greeting.
Directories

The following two types of voice processing system directories can be enabled (or disabled) in the system:

- **The Voice Mailbox Directory** is a list of mailbox subscribers, their recorded names, and mailbox numbers.

- **The Automated Attendant Directory**, provided to all Auto Attendant callers, is a list of all mailbox subscribers and extension ID owners and their recorded names.

The directories can be selected in the following ways:

- **From the Voice Mail or Automated Attendant main menu**: Say “Directory” if Automatic Speech Recognition is enabled.

- **From the Voice Mail or Automated Attendant main menu**: Say “Directory” or press the button.

- **Using a Call Routing Announcement application**: Press the Digit Translation digit designated for directory access. (See page 273 for a detailed explanation of digit translation.)

- **As a Voice Mail subscriber**: Subscribers can access the directory by saying “Directory” or by pressing whenever they are prompted to enter a mailbox number.

Nonsubscribers can only use ASR to search the directories if ASR has been enabled for the directory. Subscribers can only use ASR to search the directories if ASR is enabled for the directory and the mailbox.

If a directory is disabled or empty because no names are recorded for any of the system’s mailboxes and extension IDs, callers will not hear the prompt that allows access to the directory. However, if the caller does try to access the directory if it is disabled, he or she will be instructed that the selection is invalid and returned to the initial instructions.

Recorded names are added to the directories when the owner of the mailbox or extension ID has initialized the name. If the mailbox or ID is not initialized, the directory will include only the mailbox number or extension ID number. (Group Lists are not included in either type of directory.)

The caller can use ASR (if enabled) or press dialpad buttons to enter the name. The application then plays the closest matching directory name that corresponds to the digits entered by a caller. Once the name has been played, the system will return a menu of options, including the following:

- Listening to the previous or next name in the directory
- Listening to additional information (This option is not available if the mailbox or extension ID number has been classified in the database as a “Private” number.)
- Spelling a new name
- Toggling from a last to first name search mode
- Accepting the name
Locating a Name

Two methods may be used to enter a name using your endpoint’s dialpad digits: Quick Spell and Exact Spell. Outside callers and endpoint users with nondisplay endpoints and single line endpoints use Quick Spell. Display endpoints use the Exact Spell method. See Table 39.

- Quick Spell callers press a single digit (or button) from their endpoint dialpad for each letter or character entered. For example, dialpad button 2 shows ABC, button 3 shows DEF, etc. To enter JONES, you would press 5 6 6 3 3 7 7 7 7. Some characters are not shown on the buttons: for “Q” press 7, for “Z” press 9, for punctuation marks press the 1 button.

- Exact Spell callers press the dialpad buttons to enter the name. The number of times a button is pressed determines which character is entered, as shown on the chart below. When adjoining characters are located under the same button, press # once to advance to the next character. For example, 5 6 6 6 FWD 6 6 3 7 7 7 7 would enter “JONES.”

Table 39. Quick Spell and Exact Spell Dialpad Entries

<table>
<thead>
<tr>
<th>Number of Times Button is Pressed</th>
<th>Button 1</th>
<th>2</th>
<th>3</th>
<th>4</th>
<th>5</th>
</tr>
</thead>
<tbody>
<tr>
<td>1</td>
<td>- &amp; ( )</td>
<td>1</td>
<td></td>
<td></td>
<td></td>
</tr>
<tr>
<td>2</td>
<td>A B C '</td>
<td>2</td>
<td></td>
<td></td>
<td></td>
</tr>
<tr>
<td>3</td>
<td>D E F !</td>
<td>3</td>
<td></td>
<td></td>
<td></td>
</tr>
<tr>
<td>4</td>
<td>G H I *</td>
<td>4</td>
<td></td>
<td></td>
<td></td>
</tr>
<tr>
<td>5</td>
<td>J K L #</td>
<td>5</td>
<td></td>
<td></td>
<td></td>
</tr>
<tr>
<td>6</td>
<td>M N O Ñ</td>
<td>6</td>
<td></td>
<td></td>
<td></td>
</tr>
<tr>
<td>7</td>
<td>P Q R S</td>
<td>7</td>
<td></td>
<td></td>
<td></td>
</tr>
<tr>
<td>8</td>
<td>T U V ?</td>
<td>8</td>
<td></td>
<td></td>
<td></td>
</tr>
<tr>
<td>9</td>
<td>W X Y Z</td>
<td>9</td>
<td></td>
<td></td>
<td></td>
</tr>
<tr>
<td>0</td>
<td>@ : . ,</td>
<td>0</td>
<td></td>
<td></td>
<td></td>
</tr>
</tbody>
</table>

After the digits have been entered, the caller presses # to begin the search. The Automated Attendant application plays the name that most closely matches the digit(s) that were entered. If # is pressed without entering any digits, the caller will hear the first name in the directory.

If a user presses 0 at any time while spelling a name, the system will play a helpful prompt instructing the caller how to enter a name. (Display endpoint users will receive Exact Spell instructions; all other users will hear Quick Spell instructions.)

Changing the First/Last Name Search

Each directory can be programmed to be organized by last name or by first name. Callers accessing a directory receive system voice prompts that ask them to enter the first or last name of the person for whom they want to leave a message.

If an Executive Display, Professional Display, or Model 8560 phone with menu buttons is used to access a directory, the caller may switch back and forth from looking up a last name, to a first name simply by pressing the corresponding menu button. Callers using other Inter-Tel phones or single line phones can press 5 to toggle between last and first name.
Listening to the Next/Previous Name

After a name has been played, the caller can listen to the previous name in the directory by pressing 1. To listen to the next name, the caller can press 3.

The directory lists are circular. That is, when the end of the list is reached, the next name played will be the first name in the directory. Or, if the caller scrolls to the beginning of the list, the “previous” name played will be the last name in the directory.

Accepting a Name

When the caller is using the Voice Mail feature and accepts a name, the caller is transferred to the corresponding mailbox. Then the caller will hear either the subscriber’s recorded greeting, recorded directory name, or, if no recording has been made, the “mailbox number XXX is not available” prompt. The caller can then record a message.

If the caller is using the Automated Attendant feature, and accepts the name, the caller is transferred to the selected destination (endpoint or mailbox) if it is available. If an endpoint extension is dialed and the destination is not available, the caller is sent to the associated mailbox, if one exists.

Requesting Additional Information

After a name is selected from the directory, the caller has the option of requesting additional information. When requested, the mailbox number, extension number, and name are played (if they have not been disallowed).

Individual mailboxes and extension IDs can be programmed as “Private” in DB Programming. A Private number is included in the directory and can be dialed by a caller who knows the number, but the associated mailbox and extension number information are not available to the caller; only the name is played.

Mailboxes and extension IDs can also be programmed as “Unlisted.” That means that the number will not be included in the directory, but can be dialed if the caller knows the extension number.

Searching the Voice Mail Directory

To search the voice mail directory by saying the name:

If Automatic Speech Recognition (ASR) is enabled, you are given the option to search for a name in the directory by saying it. When prompted, say the full name (most accurate results), first name, or last name of the person you are searching for. The system searches the voice mail directory and returns a list of names. If ASR is not enabled on your voice mail system, you can use the dialpad buttons to search the directory (see below).

| NOTE | If one to nine matches are located, you can browse the list of returned names and select the correct name from the list. If the system locates 10 or more matches, you are prompted to say “First” or “Last” to limit your search to a first or last name. |
To search the directory using the endpoint’s dialpad digits:

1. Dial the Voice Mail access number. You hear the recorded system greeting.
3. The buttons on your endpoint are labeled with letters. Use Quick Spell (non-display endpoint or single line) or Exact Spell (display endpoint) to enter the name you want, as described on page 276, then press \( \text{#} \). (If you want to exit from the directory without selecting a name, press \( \text{#} \) or hang up.)
4. The Voice Mail application will play the selected name. Do one of the following:
   - Say “Accept” or press \( \text{#} \) to accept a name.
   - Say “Previous” or press \( \text{1} \) to hear the previous name in the directory.
   - Say “More” or press \( \text{2} \) to hear additional information for the selected name (if allowed).
   - Say “Next” or press \( \text{3} \) to hear the next name in the directory.
   - Say “New” or press \( \text{4} \) and say or spell a new name to search for a different name.
   - Say “Cancel” or press \( \text{#} \) to exit the directory.

**EM Directories**

Directories contain a list of extensions or mailboxes and the names programmed for them. The following directories can be enabled (or disabled) in the system:

- **The Mailbox Directory** is a list of mailbox subscribers, their recorded names, and mailbox numbers.
- **The Automated Attendant Directory** is a list of all mailbox subscribers and extension ID owners and their recorded names.

Group Lists are not included in either type of EM directory.

The directory function is accessible to all incoming callers and voice mail subscribers. The directory is built automatically through the voice mail subscribers and extensions. Recorded names are added to the directories when the mailbox or extension ID owner has initialized the name. The directory also includes extensions that do not have a voice mailbox, if the extension is programmed with an Extension ID. If the mailbox or ID is not initialized, the directory will include only the mailbox number or extension ID number.

The ASR feature allows callers to search the directory and make calls to endpoints more efficiently. If ASR is disabled, callers must use their dialpad to spell the name of the endpoint user, department, or group that they were searching for. When ASR is enabled, callers can search the directory by saying the name of the endpoint user and the directory will search and locate a match.

**Private Numbers:** Individual mailboxes and extension IDs can be programmed as “Private” in DB Studio. A Private number is included in the directory and can be dialed by a caller who knows the number, but the associated mailbox and extension number information are not available to the caller; only the name is played.

**Unlisted Numbers:** Mailboxes and extension IDs can also be programmed as “Unlisted.” That means that the number will not be included in the directory but can be dialed if the caller knows the extension number.
**Directory Access:** Callers can access the directory in the following ways:

- From the voice mail or Automated Attendant main menu, do one of the following:
  - Say “Directory”
  - OR, press \[\]
  - OR, press the **DIRECTORY** menu button

- From a Call Routing Announcement application, say or press the digit assigned through Digit Translation for directory access. For a detailed explanation of digit translation, see page 273.

If a directory is disabled or empty because no names are recorded for any of the system’s mailboxes and extension IDs, callers will not hear the prompt that allows access to the directory. If the caller presses \[\] or a Directory menu button anyway, the prompts states that the selection is invalid and the initial instructions are repeated.

Searches: With the voice mail directory, subscribers and non-subscribers can use speech recognition (if ASR is enabled) or dialpad digits (if ASR is disabled) to search for names in the directory. Subscriber names are automatically added to the directory when the user’s mailbox is initialized. ASR may not be able to locate names that are pronounced differently than they are spelled. For these types of names, use the alternate search method of spelling the name using the endpoint dialpad digits. See Table 40 on page 281.

A caller begins a directory search by selecting the directory option from the menu. The caller is prompted to say or spell the last name of the mailbox user they would like to locate. When the caller searches for a name, the directory locates all matches (first and last names) for the name. For example, if the caller searches for the name John, the system locates all of the subscribers in the system with the first or last name of John. The list of matches might include the names John Smith, John Jones, John Nielson, David John, Robert John, etc.

When one to 10 matches are located, the caller can browse the list of returned names by selecting the Previous (say “Previous” or press \[\]) and Next (say “Next” or press \[\]) options. The directory lists are circular. That is, when the end of the list is reached, the next name played will be the first name in the directory. When the correct name is played the caller says “Yes” or presses \[\] to select it. If the caller is a subscriber using a display endpoint, the names matching the search are displayed, and the caller can select a name by pressing the corresponding menu button.

If the system locates more than 10 matches, the caller can either browse the list using the options mentioned above, or filter the list by responding to the filtering questions generated by the system. To narrow the search, the system prompts the caller to indicate whether the name is the first or last name. After the caller has responded, the system returns a list of names that corresponds to the caller’s filter parameters. If the system cannot locate an exact match, it returns the closest match found.

**Additional Information:** After the caller accepts a name from the directory, he or she has the option of requesting additional information by saying “More” or by pressing \[\]. When requested, the system plays the mailbox number, extension number, and name (if they have not been disallowed).
To perform a directory search:

1. Dial the voice mail number.

Say the last name of the person you are searching for or press the appropriate dialpad buttons to spell the name. See Table 40 on page 281.

If 10 or fewer matches are located, you can access and browse the list immediately (see step 3).

If the list contains more than 10 matches, you have the option to browse or filter the list. (skip step 3, and see step 4).

3. If 10 or fewer matches are located, you have the following options:
   - Say “Yes” or press 1 to accept the name that is played.
   - Say “Previous” or press 1 to browse to the previous name in the list.
   - Say “More” or press 2 to hear more information about the subscriber.
   - Say “Next” or press 3 to browse to the next name in the list.
   - Say “New” or press 4 to search for a new name.
   - Say “Cancel” or press 4 to exit the directory.

4. If more than 10 matches are located, the system prompts you to browse or filter the list. Do one of the following:
   - Say “Yes” or press 1 to browse the list. (See step 3 for options).
   - Say “No” or press 2 to select the filter option and then respond to one of the following filter questions:
     - The system begins by asking you if the name you are searching for is the last name. If it is, say “Yes,” press 1, say the last name, or spell the last name.
     - If you did not respond to the first question, the system asks if the name you are searching for is the first name. If it is, say “Yes,” press 1, say the first name, or spell the first name.

After the list has been filtered by first or last name, a smaller list of names is returned. You can then browse the list and select the appropriate name (see step 3).
To use the dialpad buttons to spell a name: When prompted, press the appropriate dialpad buttons to spell the name. The number of times a button is pressed determines which character is entered, as shown in the digit map table below. When adjoining characters are under the same button, press **FWD** to advance to the next character. For example, 5666 **FWD** 6637777 enters “Jones.”

**Table 40. Dialpad Alphanumeric Coding Layout**

<table>
<thead>
<tr>
<th>Button</th>
<th>1</th>
<th>2</th>
<th>3</th>
<th>4</th>
<th>5</th>
</tr>
</thead>
<tbody>
<tr>
<td>1</td>
<td>-</td>
<td>&amp;</td>
<td>(</td>
<td>)</td>
<td>1</td>
</tr>
<tr>
<td>2</td>
<td>A</td>
<td>B</td>
<td>C</td>
<td>'</td>
<td>2</td>
</tr>
<tr>
<td>3</td>
<td>D</td>
<td>E</td>
<td>F</td>
<td>!</td>
<td>3</td>
</tr>
<tr>
<td>4</td>
<td>G</td>
<td>H</td>
<td>I</td>
<td>*</td>
<td>4</td>
</tr>
<tr>
<td>5</td>
<td>J</td>
<td>K</td>
<td>L</td>
<td>#</td>
<td>5</td>
</tr>
<tr>
<td>6</td>
<td>M</td>
<td>N</td>
<td>O</td>
<td>ñ</td>
<td>6</td>
</tr>
<tr>
<td>7</td>
<td>P</td>
<td>Q</td>
<td>R</td>
<td>S</td>
<td>7</td>
</tr>
<tr>
<td>8</td>
<td>T</td>
<td>U</td>
<td>V</td>
<td>?</td>
<td>8</td>
</tr>
<tr>
<td>9</td>
<td>W</td>
<td>X</td>
<td>Y</td>
<td>Z</td>
<td>9</td>
</tr>
<tr>
<td>0</td>
<td>@</td>
<td>:</td>
<td>.</td>
<td>,</td>
<td>0</td>
</tr>
</tbody>
</table>

**Outside caller search method:** Outside callers must use the “Quick Spell” method of searching the directory, if ASR is not enabled. When an outside caller accesses the voice mail directory, he or she is prompted to press the digits on the dialpad that correspond to the name they are searching for. For example, dialpad button **2** corresponds to ABC, button **3** corresponds to DEF, etc. To enter the name “Jones,” the caller would press **5 6 6 3 7 7 7**. The system finds the closest match based on the caller’s entry.

Press **7** for “Q,” **9** for “Z,” and **1** for punctuation marks.

When the caller accepts the directory name from voice mail, the caller is transferred to the corresponding mailbox. After the transfer the caller either hears the subscriber’s recorded greeting, their recorded directory name, or, if no recording has been made, the “mailbox number XXX is not available” prompt. The caller can then record a message.

When the caller accepts the directory name from the Automated Attendant, the caller is transferred to the selected destination (endpoint or mailbox) if it is available. If the destination is not available, the caller is transferred to the associated mailbox, if one exists.
E-mail Reader

The EM voice processing system supports E-mail Reader which, if enabled, provides access to voice mail, e-mail, and fax messages that are located in an advanced mailbox. E-mail Reader uses Text-to-Speech (TTS) capabilities to convert the text in an e-mail message to a .wav file.

When the user accesses an e-mail message, E-Mail Reader announces the e-mail according to the envelope options that the user set. Then, after the envelope is played, E-mail Reader reads the text in the body of the message. Additional options for e-mail messages include reply, forward, save, and delete. E-mail Reader is supported only on EM voice processing systems.

When the user accesses a fax message, E-mail Reader announces the fax envelope according to the envelope options that the user set up. Because a fax message is received by the e-mail server as an attachment to an e-mail message in the form of a .tif file, E-mail Reader cannot read the text in the body of the fax. To view a fax and read it, users must access it from their e-mail account. Fax options include saving, deleting, and forwarding to another fax machine.

For detailed information about using standard voice mailboxes and advanced mailboxes with E-mail Reader capabilities, see the appropriate endpoint user guide.

Fax-On-Demand

The Fax-On-Demand feature provides fax services to callers. It is a specially programmed Call Routing Announcement application that uses digit translation to allow callers to select the documents they want to have faxed to them.

With Fax-On-Demand, callers can use a DTMF endpoint to request one or more documents from the company’s fax library, the library that is also used in the Unified Messaging faxing feature. When the request is completed, the voice processing system places a call to the caller’s fax machine to deliver the requested documents.

The Fax-On-Demand process is shown in the following steps:

1. The caller reaches the Call Routing Announcement application either through direct ring-in or transfer.

2. After listening to the introductory recording, the caller selects a document by entering a document number or using digit translation, depending on digit translation node programming. If the document is selected by number, the system will ask the caller to verify the entry by pressing #. If the number is invalid, the system alerts the caller and returns to the introductory recording.

3. The system checks the availability of the document, and one of the following occurs:
   - If the document is available, the system tells the caller how many documents have been selected and how many more documents can be requested (as determined by the Maximum Selections Allowed value in the database). It then returns to the recording to allow the caller to select more documents or end the selection process.
   - If the document has been deleted or is being updated, the caller is told that the document is unavailable and returns to the recording.

4. If necessary, the caller selects additional documents, and the system checks for their availability as described above. If the maximum number of allowed documents has been reached, the system informs the caller and gives the option of returning to the greeting or ending the selection process. If a caller selects the same document more than once, the system informs the caller of the error and allows another selection.
5. If appropriate, the caller can cancel the document selections at this point by pressing the Cancel Fax Selections digit (usually *). However, once the next steps are completed, the faxes will be delivered.

6. When the caller ends the selection process by pressing the End Fax Selections digit (usually #), or reaches the allowed maximum number of documents, the system requests fax dialing information from the caller.

7. The caller enters and verifies his or her extension number (to be printed on the cover sheet) and the caller’s fax machine phone number.

8. The system checks the entered number, and one of the following occurs:
   - If the number is valid, the system tells the caller when the document will be delivered (this depends on the programmed fax delivery times in the voice processing system database) and returns to the recording.
   - If the number cannot be dialed by the Message Notification/Retrieval application due to toll restrictions, the system tells the caller that the number cannot be dialed and asks for a new number.
   - If the caller enters an invalid number, the system tells the caller the number is invalid and asks for a new number.

9. The caller hangs up or returns to step 2 on page 282 to select more documents. (If more documents are selected and canceled, the documents that were selected and accepted previously will still be sent. Each time the caller accepts the documents and enters the dialing information, the documents are prepared for transmittal. They cannot be canceled after that point.)

10. The Message Notification/Retrieval application places a call to the caller’s fax machine as soon as a fax port is available, unless one of the following occurs:
   - If the fax was requested outside of the programmed fax delivery times, the system will wait until the Fax Delivery Start Time on an allowed day of the week before it attempts to send the document. The Fax Delivery Start and Stop times determine when faxes can be sent. Start and stop times can be set to any time period, up to 24 hours. The programmer can also determine which days of the week faxes will be sent.
   - If other calls are going out and the programmed Maximum Number of Outgoing Calls has been met, the system waits for an available outgoing call to send the fax.
   - If a fax port or trunk is not available or if a requested document is not available, the system delays delivery until the resource or document is available.
   - If a requested document has been deleted, the system will complete the delivery, but will indicate on the cover sheet that the document is no longer available.
   - If a fax delivery fails after one or more pages have been transmitted, the system remembers which pages were not sent and tries again to send only those pages. For example, if the transmission failed in the middle of page 3 of 5, the system would attempt to send only pages 3, 4, and 5 on the retry. The cover sheet on the retry will say, “This fax is the continuation of...” and the description for the fax will include “continued” in the title.

Documents are stored in the fax library using the system administrator’s mailbox. To do so, a fax machine places a call to the system administrator’s mailbox and sends the document. For instructions, see “Importing Fax Documents” on page 70.
Fax-On-Demand transmits a cover sheet with each faxed document that includes the recipient’s extension number, your company name (as programmed in the database), the date, the number of pages to be sent, and a list of included documents. If you want, you can store a company logo to be used on the cover sheet. The logo can be up to 5.5 inches high; if it is larger, the system automatically cuts it off at 5.5 inches. The logo document is stored, like other fax documents, using the system administrator mailbox.

If you have several fax documents, you may want to create and store a “catalog” document that callers can request before continuing their selections. For example, the catalog can include the document number, title, description of the contents, number of pages, and last revision date for each document.

Fax documents can be saved and restored using the Voice Processing Save And Restore database program. The system Save And Restore operation has no effect on fax documents. However, if the system database is defaulted, all fax documents are deleted.

The digit translation node used for fax document selection should have a greeting that explains how to select documents. For example, it can say something like, “Welcome to Fax-On-Demand. For a product list, press 1. For a price list, press 2. When you have made your selections, press # to continue. Or, to cancel your selections, press *.” Or, you can use a series of digit translation nodes that break the document selection into categories and/or allow callers to dial individual document numbers. If fax delivery times are set to specific days or times, the introductory recording should also include this information.

**Record-A-Call**

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**IMPORTANT** The Record-A-Call feature is required to use Record-A-Call.

If the voice processing system has a Record-A-Call application, the users can enter a feature code to record ongoing calls in their designated Record-A-Call mailboxes. Users can retrieve the recorded messages later, just as they would any other mailbox message.

The Record-A-Call mailbox records the call as a Voice Mail message. All parties will be included in the recording. If appropriate, the Record-A-Call mailbox can be programmed to play a message announcing that the Record-A-Call feature is in progress. Separate messages can be played for day and night modes.

A Record-A-Call tone can be programmed to alert callers at the beginning of the recording. Also, it can be programmed to beep periodically throughout the recording.

There are two ways an endpoint can be programmed to use this feature:

- The endpoint can use its personal mailbox, or any other mailbox, as the assigned Record-A-Call mailbox. No other mailbox can be selected. (This is the default programming for all endpoints.)

- The endpoint can be programmed with a default mailbox, but with the option of selecting a different mailbox. If the user chooses to not enter a mailbox number, the system automatically selects the default mailbox.

When the requesting endpoint user hangs up, all parties are automatically disconnected. If all parties on the call hang up, except the requesting endpoint user, an intercom call will remain connected between the requesting endpoint and the Record-A-Call application. This allows the user to make additional comments before ending the recording.

When the endpoint user turns off Record-A-Call or ends the call being recorded, the voice processing system delivers the message to the mailbox. The endpoint associated with the mailbox will receive message waiting indications as usual. For message retrieval instructions, refer to the appropriate Inter-Tel endpoint user guide.
The service technician can set a maximum length for Record-A-Call messages. The Record-A-Call Maximum Message Length timer can be set at 0–600 minutes. A 0 setting allows messages of any length, limited only by the available disk space. The default is 30 minutes.

In a network setting, the Record-A-Call destination does not have to be on the same node as the endpoint, but the voice processing system must have a mailbox for that endpoint.

For more information on using the Record-A-Call feature, see page 200.

**Scheduled Time-based Application Routing (STAR)**

Scheduled Time-based Application Routing (STAR) allows you to have applications with alternate greetings and/or different programming set up for holidays, weekends, and other scheduled events.

A STAR application is basically a “routing table” for voice processing system applications. When a direct ring-in call (from a trunk group or call routing table) rings in to a STAR application, it sends the call to another application, according to its programmed schedule. The caller will not be aware of this transfer, but will hear the programmed day or night greeting for the destination application. The STAR application itself does not play a greeting.

STAR can be used with any type of application except Auto Attendant Recall and Record-A-Call. You can even send calls from one STAR application to another, thereby “chaining” the applications to increase the number of available schedules.

A STAR application can contain up to 20 scheduling entries with the following fields:

- **Application**: This is the application (Call Routing Announcement, Automated Attendant, etc.) that is used when the scheduling information applies to the incoming call.

- **Start/Stop Date**: If the schedule is going to be active on a single day or for a period of days, it will have Start and Stop Dates. (To have the schedule active on only one day, the Start and Stop Dates can be the same day.)

- **Days of the Week**: The schedule can be used on specific days of the week.

- **Specific Times or Day/Night Mode**: The schedule can be set to be active for a specific period of time on the selected day(s). Or, it can be set to be active in day or night mode.
Table 41 shows several samples of STAR entries. Note that there are two entries for Labor Day and Memorial Day, and that Christmas Day and Christmas Holiday overlap.

**Table 41. STAR Programming Entries**

<table>
<thead>
<tr>
<th>Entry</th>
<th>Description</th>
<th>Date</th>
<th>Day</th>
<th>Time Interval</th>
<th>Day/Night</th>
<th>Application</th>
</tr>
</thead>
<tbody>
<tr>
<td>1</td>
<td>Labor Day</td>
<td>Sep 1</td>
<td></td>
<td>10:00 AM–2:00 PM</td>
<td></td>
<td>2600</td>
</tr>
<tr>
<td>2</td>
<td>Labor Day</td>
<td>Sep 1</td>
<td></td>
<td></td>
<td>Both</td>
<td>2601</td>
</tr>
<tr>
<td>3</td>
<td>Memorial Day</td>
<td>May 30</td>
<td></td>
<td>8:00 AM–5:00 PM</td>
<td></td>
<td>2605</td>
</tr>
<tr>
<td>4</td>
<td>Memorial Day</td>
<td>May 30</td>
<td></td>
<td>5:00 PM–8:00 AM</td>
<td></td>
<td>2606</td>
</tr>
<tr>
<td>5</td>
<td>Christmas Day</td>
<td>Dec 25</td>
<td></td>
<td></td>
<td>Both</td>
<td>2607</td>
</tr>
<tr>
<td>6</td>
<td>Christmas Holiday</td>
<td>Dec 24–Jan 2</td>
<td></td>
<td>Both</td>
<td>2608</td>
<td></td>
</tr>
<tr>
<td>7</td>
<td>Weekends</td>
<td>Sat, Sun</td>
<td>Both</td>
<td></td>
<td></td>
<td>2612</td>
</tr>
<tr>
<td>8</td>
<td>Lunch Break</td>
<td>M–F</td>
<td></td>
<td>11:00 AM–1:00 PM</td>
<td>Night</td>
<td>2613</td>
</tr>
<tr>
<td>9</td>
<td>Night Shift</td>
<td>M–F</td>
<td></td>
<td></td>
<td>Day</td>
<td>2614</td>
</tr>
<tr>
<td>10</td>
<td>Day Shift</td>
<td>M–F</td>
<td></td>
<td></td>
<td></td>
<td>2615</td>
</tr>
</tbody>
</table>

The first entry in the table is checked first, and so on, until a match is found. Therefore, when entries occur on the same date or day but at different time intervals, it is important to program the applications in the appropriate sequence.

For example:

- The Labor Day entry, with a specific time, comes before the Day/Night entry so that a different message (2600 in this case) is played only from 10:00 AM to 2:00 PM.
- The early time intervals come before the later intervals on Memorial Day.
- Dec. 25 comes before the range of dates Dec. 24–Jan. 4 so that the special holiday message will be played on that day only.

Once a match is found, the application corresponding to the matched entry is invoked. If no match is found, the application programmed in the “Default Application” field is used.

When programming the days and times, be careful not to overlap. For example, if you place a date entry for Dec. 25 after a day of the week entry for Monday, and Christmas falls on a Monday, the “Monday” application will be selected instead of the “Dec. 25” application.

**SMDR Information Storage and Retrieval**

An external Voice Processing Unit (VPU) can be used to capture Station Message Detail Recording (SMDR) information. SMDR information is available only through an IP connection on a Voice Processing Unit running Windows 2000. It is not available in basic voice mail or through Enterprise Messaging. Also, if behind a firewall, a port must be open to allow transfer of SMDR information to the VPU (or other PC set up to buffer SMDR). The SMDR information is stored in a file on the VPU hard drive for later retrieval.

Every time a call is completed, the SMDR feature records the call, formats it according to SMDR programming (See “Station Message Detail Recording” on page 250), and sends the record to the SMDR output device, if enabled. The formatted record is also sent to the voice processing system if SMDR buffering is enabled.

EM does not provide support to buffer SMDR.
The voice processing system PC adds each call record to the SMDR buffer file as it is received. The amount of disk space allocated for SMDR storage is determined in DB Programming. The default value is zero megabytes of memory. If the call records use disk space exceeding 80% of the allotment, the system generates an alarm and displays it at the primary Attendant station and all Administrator stations. Another alarm message is generated if the allotted space is 100% filled. If the maximum buffer capacity is exceeded, the oldest call record(s) are discarded as new ones arrive.

Because each SMDR call record takes up approximately 81 bytes on the voice processing system PC, nearly 13,000 calls can be stored in each megabyte of hard disk space.

The SMDR records, which are stored in ASCII/binary format, are buffered until they are cleared by the programmer. If necessary, the programmer may save and/or restore SMDR data (along with the system database) to floppy disks using the Voice Processing Save and Restore feature.

In a network setting, the voice processing system will buffer SMDR records that are sent from other nodes in the network.

**Basic Voice Mail**

Inter-Tel 5000 software includes licensing to support four BVM ports. Additional ports can be added with the purchase of licenses that support four ports each. Although each BVM port must be licensed, single-port licenses are not available. The Inter-Tel CS-5200 and CS-5400 systems support up to eight BVM ports, and the Inter-Tel CS-5600 supports up to 16 BVM ports. As needed and licensed, BVM ports are allocated dynamically. BVM storage capacities by type of system appear in Table 47 on page 305.

Before an external voice processing system such as Enterprise Messaging can be attached to any of the Inter-Tel 5000 systems, BVM must be disabled in DB Programming, which makes the feature unavailable for use.
Voice Mail

The Voice Mail feature allows callers to send and receive recorded messages from any endpoint or DTMF endpoint. (Dial pulse signaling is not recognized by the voice processing system. Callers using dial pulse endpoints will be routed to the Voice Mail application’s Dial-0 Destination for assistance.)

Voice Mail and Message Notification/Retrieval Applications

Two types of applications are combined to provide the Voice Mail feature: Voice Mail and Message Notification/Retrieval.

- **Voice Mail**: This application handles all calls that are directed to Voice Mail (other than the Message Notification/Retrieval application) placed by subscribers and nonsubscribers. Callers will hear the main company greeting, followed by a menu of available options. Endpoints can forward or transfer calls directly to their mailbox using this application’s extension number. This application can also be the message center for the subscribers’ endpoints.

- **Message Notification/Retrieval**: Only one Message Notification/Retrieval application may be created for the system. When endpoint users respond to message indications left by Voice Mail, they reach the Message notification/Retrieval application and are prompted only for their password.

Endpoint users and outside callers gain access to the Voice Mail application in the following ways:

- By dialing the phone number or extension number assigned to the Voice Mail application
- By being transferred to Voice Mail by Automated Attendant
- By being forwarded to the Voice Mail when a called endpoint is unavailable (via the Call Forwarding or System Forwarding features)
- By responding to Message Waiting indications

Callers will hear the Voice Mail company greeting and recorded instructions that tell them what to do next. Users listen to the prompts and press the dialpad button that corresponds to the choice they want. If the user does not respond immediately, a second set of prompts is played. Most prompts are interruptible, and users can press the applicable button at any time during the prompt. The prompt will then stop and the system will act on the requested choice.
Mailboxes

A mailbox is a storage location on the voice processing system PC hard disk that stores all messages that have been directed to it. (The hard disk also stores prompts, greetings, and special programming.) Each “Subscriber” (member of the voice message system) is assigned a unique mailbox number. At system default, the passwords are the same as the mailbox numbers. (For example, the default password for mailbox 1001 is “1001.”)

There are three types of mailboxes on the Voice Mail system: Standard, Receive-Only and System Administrator. There can also be Extension IDs which are endpoints that do not have mailboxes, but that are included in the recorded directory.

- The Standard mailbox owner receives all of the available subscriber options as authorized through DB Programming. Standard mailbox features include:
  - Listening to new messages
  - Recording and sending messages
  - Deleting and restoring messages
  - Saving messages
  - Personal options (greeting, password, etc.)
  - Remote messaging (if authorized through DB Programming)

- A Receive-Only mailbox type may only listen to new messages, save, delete, or recover them and access personal options. This type of mailbox is restricted from recording and sending messages and has no access to remote messaging.

- There can be only one Voice Mail Administrator. The Voice Mail Administrator has all of the options a Standard mailbox subscriber has, plus these features:
  - Record and send broadcast messages to all subscribers of the Voice Mail system
  - Provide mailbox and group list maintenance
  - Record and select Custom Audiotex Recordings

- Extension IDs are used in conjunction with transferring calls through the Automated Attendant or using the Extension Directory. The extension ID allows callers to be transferred to endpoint extensions which do not have mailboxes and allows the endpoint user to record a directory name and set up a password.

Calls transferred from the Automated Attendant or a Call Routing Announcement application to an endpoint with a mailbox can be screened, announced, or unannounced. Separate programming flags determine the methods used for transferring calls to endpoints with mailboxes.

In DB Programming, each mailbox is assigned a dial-0 operator destination. When a caller presses 0 while listening to the mailbox’s personal greeting or recording instructions, or while recording a non-subscriber message, the operator destination is called. The operator can be one of the following types:

- **Mailbox:** If the operator is a mailbox, the caller will hear the operator destination’s personal greeting after pressing 0.

- **Endpoint extension number:** If the operator destination is an extension number, the caller will hear the Automated Attendant transfer prompt while the call is placed to the extension number (“Please hold while I transfer your call to...”), unless it has been disabled at the mailbox. If the endpoint user does not answer, the caller will be asked if he wants to leave a message, if the operator’s extension has an associated mailbox. If the caller chooses not to leave a message, the call returns to the main menu of the application being used (Call Routing Announcement, Voice Mail, or Automated Attendant).
- **Application extension number**: If the operator destination is an application’s extension number, the call is transferred to that application’s main menu.

- **Operator**: If the operator destination is set to “Operator,” the caller will be transferred to the operator destination programmed in the voice processing system database.

- **None**: If the operator destination is set to “None,” the 0 will be ignored.

### Group Lists

Group lists are programmed in the database. They are lists of mailboxes that can be used by any Subscriber for sending messages to several mailboxes simultaneously. The information that is programmed for group lists include the following:

- Group list description
- Group list number
- Mailboxes included in the list

There can be up to 1000 group lists in the system and up to 1500 entries per group list. Group Lists are not included in the Automated Attendant or Voice Mail directories. Instructions for using group lists are provided in the appropriate endpoint user guides.

### Message Notification to Endpoints

Each mailbox has a programmed Message Notification Endpoint. This is usually the extension number that matches the mailbox number (associated mailbox). However, a different number can be used for the mailbox’s message notification endpoint (non-associated mailbox).

- **Associated**: When the extension number, message notification number, and mailbox number for an endpoint all match, that endpoint user can forward calls to Voice Mail and have the call go directly to the associated mailbox’s personal greeting. (Even if other mailboxes use that extension number as their message notification extension.)

- **Non-Associated**: The extension number assigned as the Message Notification Endpoint for a non-associated endpoint is an extension number that does not match the mailbox number. (For example, a hunt group pilot number can have a mailbox, but the message notification must be sent to a specific endpoint so that a message lamp can be lit or message notification signal can be sent.) If an endpoint user’s extension number does not match a mailbox number, and the endpoint user forwards calls to Voice Mail, the caller will hear the main Voice Mail greeting and must enter a mailbox number.

When a mailbox receives a message, the message notification endpoint is signaled. If it is an Inter-Tel endpoint, the  button is lit and the display shows that a message has been received. A single line endpoint will receive message waiting signals, if enabled system-wide.
Cascading Remote Message Notification

Remote Messaging is a subscriber feature that is enabled in DB Programming. Subscribers may program a series of specific phone numbers, a “cascade,” for the Voice Mail system to call when new messages are received by their mailboxes. Refer to the appropriate user guides for instructions on using cascading remote message notification.

Primary and Alternate Notification

Each mailbox can have a Primary and/or an Alternate notification cascade of numbers. The notification numbers are programmed to be accessed during specific days and times.

The Voice Mail system monitors subscriber mailboxes continuously. During monitoring, if new messages have been received, the system will check to see if Primary notification is turned on. If it is, the system will check the day and time programming. If the current time is within the programmed notification day and time, the Voice Mail system will make the notification call(s) to the numbers in the Primary Notification cascade. If Primary Notification is turned off or if the day or time does not match, the Voice Mail system will immediately make the same check for Alternate Notification and possibly place calls to the numbers in the Alternate Notification cascade.

When the voice processing system makes a notification call, it moves through the appropriate notification cascade level by level as determined by the notification and retry programming described below. When the system reaches the last level in the cascade, it will return to level 1 and begin again, if necessary.

If the system is not able to place the call within the parameters of the programmed notification day and time (due to busy facilities or invalid programming), the mailbox will receive a message stating that notification could not be completed (the mailbox owner will hear the message next time he or she accesses the mailbox.)

The number of calls that the voice processing system can process simultaneously is determined by the Maximum Number of Outgoing Calls flag. It defaults to 2, but can be programmed to use all of the enabled voice processing system voice channels.

Message Flags

When the mailbox is being programmed for remote notification, message notification can be set to place the notification call when any message is received or only when priority messages are received. (This defaults to “All Messages.”) Another mailbox flag, called “Call For Each New Message,” determines when the voice processing system will attempt to make the notification call when more than one message is waiting to be heard:

- If the Call For Each New Message flag is enabled, the voice processing system will attempt immediate notification each time a new message (or priority message, depending on the option selected) is received, regardless of the status of the applicable retry timer. If a retry is in progress, it will place the call to the number on the cascade table that would have received the retry call, as described below. (For example, if the voice processing system was on level 3 of a cascade, attempting retries, it will not start over again at level 1. It will place the call to level 3.)

- If the Call For Each New Message flag is disabled, new messages will not generate a remote notification attempt until the Subscriber has logged on and listened to the waiting message(s). (However, retry calls will be attempted if the applicable retry timer expires, as described below.)
Retry Call Attempts

The mailbox can be programmed to retry notification calls when it encounters a busy trunk or trunk group. Each cascade level can be programmed with a Number of Call Attempts value of 1–1000 calls.

If the mailbox user accesses the mailbox between the time the message is received and when remote notification is successful, the system will stop attempting remote notification. It is assumed that the mailbox user listened to the message when the mailbox was accessed. However, the [MSG] button at the endpoint will remain lit if there are any messages that have not been heard.

Each level also has a Number Called Busy timer and a Pager Notification Retry timer or Personal Number No Answer timer. These timers determine how long the voice processing system will wait before making the next notification attempt when messages are waiting to be heard (unless it is overridden by the Each New Message flag). These timers can be set to a value of 0–255 minutes. If set to 0, it retries the cascade level immediately, for as many attempts as allowed by the Number of Call Attempts field. If the message(s) has not been picked up by then, the system moves to the next cascade level. The default setting for the Pager Notification Retry timer is 20 minutes. The Number Called Busy and Personal Number No Answer timers default to 5 minutes.

Notification Numbers

Each cascade level has a programmed notification number that can be an extension number or an outside phone number.

When an outside phone number is used, the system accesses an outgoing trunk using the code programmed for that level. The code contains a pre-programmed trunk access code followed by an outgoing access termination feature code, if needed.

Notification calls to outside phone numbers can be sent to a pager or to a number where a person will answer (personal number).

- **If the call goes to a pager**, the system dials the programmed Pager Dial String, and then disconnects the call. The dial string should include digits the paging company requires when the call is answered, the pager LCD number, and the pager termination code, if needed. Also, if appropriate, the pager string can be programmed to show the number of the mailbox that placed the call and/or the number of messages in the mailbox.

- **If the call goes to a personal number**, the system plays a prompt announcing that the call is from Voice Mail, the directory name or mailbox number that originated the call, and user instructions. The listener can then enter the mailbox’s password to gain access and hear the message. The listener has three chances to enter a correct password. If the system receives an incorrect password three times, the Voice Mail system disconnects the call immediately, and the attempt is considered unsuccessful.

---

**IMPORTANT**

The 0-minute timer value should be used cautiously. If the system is forced to make several calls in quick succession, it will impair the voice processing system's ability to place other outgoing calls. It should be used primarily with cascades in which all of the members want to receive notification at approximately the same time. In this case, all levels except the last could be set to 0-minute retries with 1 allowed call attempt. However, the last level in the cascade should be programmed with a longer Pager Notification Retry timer so that the system does not continuously cycle through the cascade levels. (Continuous cycling is also prevented somewhat by a per-mailbox 15-call limit. Due to FCC regulations, the voice processing system can allow a mailbox to make only 15 calls in rapid succession without a 10 minute pause.)
Subscriber Features

Before subscribers can use voice mail, they must set up their mailbox. Mailbox setup and configuration options include:

- Initializing a mailbox
- Setting up an e-mail password (if E-Mail Reader is enabled)
- Recording personal greetings
- Recording a directory name
- Changing a password
- Changing a message envelope
- Programming remote messaging
- Changing the call screening transfer method
- Programming a fax destination
- Changing the message search order

After the subscriber’s mailbox is configured, mailbox tasks include the following:

- Accessing a mailbox
- Disabling and enabling ASR (if ASR is enabled)
- Accessing voice mail messages
- Accessing e-mail, and fax messages (if E-Mail Reader is enabled)
- Recording and sending voice messages
- Canceling unheard voice messages
- Recovering deleted messages

Instructions for all of the subscriber tasks are located in the endpoint user guides. In addition to instructions for the tasks listed above endpoint user guides contain frequently asked questions, guidelines for using ASR, information about using E-Mail Reader, and flowcharts for the standard voice mailbox and the advanced mailbox.

For the appropriate user guide for voice mail and mailbox information, see “Endpoint User Guides” on page 358.
Unified Messaging

The Unified Messaging (UM) feature links the voice processing system with e-mail and/or provides outgoing fax capabilities. The Voice Processing Unit (VPU) uses Unified Messaging version 2.2. The Enterprise Messaging (EM) voice processing system uses UM version 2.3 or UM Open Standards Edition (OSE). The following sections provide information on each version.

Unified Messaging is not supported in Basic Voice Mail (BVM).

Unified Messaging allows subscribers to combine their voice mail, e-mail, and faxes in one mailbox. The Inter-Tel 5000 system supports the following versions of Unified Messaging:

- **Unified Messaging v2.2**: Allows subscribers to view, play, and manage their voice mail from their desktop PC using Microsoft Outlook®. Unified Messaging v2.2 is for Voice Processing Unit (VPU). The following three levels of Unified Messaging features are included in v2.2.
  - **Level 1 Voice Mail As E-mail Attachments**: Provides users the ability to receive a Voice Mail message as an e-mail message with a voice attachment. The voice attachment will be in the form of a “wave” (.wav) file that can be played on any computer equipped with a sound card, speakers, and multimedia software (such as Sound Recorder). Users can choose to receive the messages as Voice Mail and e-mail (Copy & Forward) or as e-mail only (Forward Only). The e-mail server can use SMTP/POP3, VIM, MAPI protocol. The number of users with Level 1 is limited by a security button attached to the voice processing system PC.
  - **Level 2 with Microsoft Outlook/Exchange**: To use this voice processing system level, the mailbox E-mail Gateway field must be programmed for Unified Messaging, and the user’s PC must have Microsoft Outlook/Exchange and the Unified Messaging Client software installed. Voice mail messages are sent to the Voice Mail server and are stored on the voice processing system. This allows the user to play the messages using the Inter-Tel endpoint speakerphone or the PC’s speakers. The number of users with Unified Messaging is limited by a security button attached to the voice processing system PC.
  - **Level 3 with Microsoft Outlook/Exchange and Outbound Fax**: This level is the same as Level 2, with outbound fax server features added. Use of the outbound fax server functionality does not require a mailbox, only the Microsoft Outlook/Exchange and Unified Messaging Client software. Therefore, the number of outbound fax users is not limited by the security button; only the number of mailboxes with Unified Messaging enabled are limited.
• **Unified Messaging version 2.3**: Allows subscribers to view, play, and manage their voice mail from their desktop PC using Microsoft Outlook. Unified Messaging v2.3 is for Enterprise Messaging. Unified Messaging v2.3 supports the following solutions:

  - *Integrated Solution*: Receives voice mail messages and faxes as e-mail attachments. This solution is set up primarily in telephone system and voice messaging system programming. The subscriber cannot use e-mail to reply to a voice mail message but instead must forward the message or send a new voice or e-mail message to the sender.
  
  - *Outlook Client-Based Solution*: Receives and sends voice mail messages and sends faxes via the Inter-Tel Enterprise Messaging system. Receives faxes via e-mail using the Microsoft Outlook Client. Subscribers can use e-mail to reply to voice mail. A built-in Unified Messaging Server in the Inter-Tel 5000 system functions as a filter and store for the Outlook Client and requires subscribers to install the Unified Messaging Client application on their PC. Most Unified Messaging v2.3 subscribers use this solution.

• **Unified Messaging Open Standards Edition (OSE)**: Allows subscribers to view, play, and manage their voice mail from their desktop PC using any Internet Message Access Protocol (IMAP) v4-compatible e-mail client, such as Microsoft Outlook, GroupWise®, or Lotus Notes. Subscribers can receive voice mail messages and receive and send faxes via e-mail. Unified Messaging OSE is a client-free application and does not require the Unified Messaging Client software application; instead, all software components are located on the Enterprise Messaging server.
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Introduction

This section contains background information on a variety of topics that may be useful for system and voice mail administrators.

Environmental Requirements

Table 42 shows the ranges of environmental conditions for operation and storage of Inter-Tel 5000 platform equipment and endpoints.

### NOTICE

**Possible Malfunction Due to Overheating.** Never exceed the maximum operating temperature listed in Table 42. When installing the chassis and endpoints, allow a sufficient margin for error in case of air conditioning failure, routine maintenance, plant shutdown, etc. As a general guideline, if the conditions are suitable for office personnel, they are also suitable for all equipment and endpoint operation. A properly controlled environment will help to extend the operating life of the equipment.

<table>
<thead>
<tr>
<th>Requirements</th>
<th>In Operation</th>
<th>In Storage</th>
</tr>
</thead>
<tbody>
<tr>
<td>Temperature</td>
<td></td>
<td></td>
</tr>
<tr>
<td>Equipment Chassis</td>
<td>32 to 104 °F (0 to 40 °C)</td>
<td>+40 to 185 °F (–40 to 85 °C)</td>
</tr>
<tr>
<td>Station Instruments</td>
<td>32 to 113 °F (0 to 45 °C)</td>
<td>–40 to 185 °F (–40 to 85 °C)</td>
</tr>
<tr>
<td>Relative Humidity</td>
<td>5–95%</td>
<td>5–95%</td>
</tr>
<tr>
<td>(Non-Condensing)</td>
<td></td>
<td></td>
</tr>
<tr>
<td>Altitude</td>
<td>Up to 10,000 ft (3048 m)</td>
<td>Up to 40,000 ft (12,192 m)</td>
</tr>
</tbody>
</table>

System Capabilities

The Inter-Tel 5000 supports the telecommunication equipment required to provide seamless integration into the Public Switched Telephone Network (PSTN) and the Internet.

Trunks and Spans

The Inter-Tel 5000 platform supports the trunks and multi-channel spans described in the following sections.

Central Office (CO) [Local Exchange] Trunks

Including the two loop start circuits built into the Inter-Tel 5000 Base Server, up to 14 loop start trunks can be supported by the Inter-Tel 5000 platform. Up to three Loop Start Modules (LSM) can be installed, and LSMS are available with either two or four ports.

T1/E1/PRI Spans

With three expansion bays, the Inter-Tel 5000 Base Server supports up to six T1, T1/PRI, or E1/PRI spans. Single-port T1/E1/PRI modules (T1M) support one span each, and Dual T1/E1/PRI modules (T1M-2) support two spans each. A T1 span supports up to 24 trunks; a T1/PRI span supports up to 23 trunks; and, an E1/PRI span supports up to 30 trunks. Trunk options in DB Programming include Loop Start, Off-Premises Extension (OPX), Direct Inward Dial (DID) [Direct Dialing Inward (DDI), in Europe], T1, T1/PRI, E1/PRI, and T1/E&M.
Capacities of All-Trunk Systems

Table 43 summarizes system trunk capacity in terms of the number of trunks provided in possible all-trunk systems. A typical installation may include a combination of hard-wire trunks and IP gateway trunks.

**Table 43. Trunking Capacities by Type of Module**

<table>
<thead>
<tr>
<th>Module Type</th>
<th>Trunk/Span Type</th>
<th>Modules</th>
<th>Ports or Channels per Module</th>
<th>Trunk Capacity</th>
<th>Built-In LS Ports</th>
<th>Total Trunks</th>
</tr>
</thead>
<tbody>
<tr>
<td>T1M</td>
<td>E1/PRI</td>
<td>3</td>
<td>30</td>
<td>90</td>
<td>2</td>
<td>92</td>
</tr>
<tr>
<td>T1M-2</td>
<td>E1/PRI</td>
<td>3</td>
<td>30</td>
<td>180</td>
<td>2</td>
<td>182</td>
</tr>
<tr>
<td>T1M</td>
<td>T1</td>
<td>3</td>
<td>24</td>
<td>72</td>
<td>2</td>
<td>74</td>
</tr>
<tr>
<td>T1M-2</td>
<td>T1</td>
<td>3</td>
<td>24</td>
<td>144</td>
<td>2</td>
<td>76</td>
</tr>
<tr>
<td>T1M</td>
<td>T1/PRI</td>
<td>3</td>
<td>23</td>
<td>69</td>
<td>2</td>
<td>71</td>
</tr>
<tr>
<td>T1M-2</td>
<td>T1/PRI</td>
<td>3</td>
<td>23</td>
<td>138</td>
<td>2</td>
<td>73</td>
</tr>
<tr>
<td>LSM-2</td>
<td>Loop Start</td>
<td>3</td>
<td>2</td>
<td>N/A</td>
<td>N/A</td>
<td>8</td>
</tr>
<tr>
<td>N/A–CS-5200</td>
<td>IP Gateway Trunks(^1)</td>
<td>0</td>
<td>N/A</td>
<td>32</td>
<td>2</td>
<td>30</td>
</tr>
<tr>
<td>N/A–CS-5400 or CS-5600</td>
<td>IP Gateway Trunks(^1)</td>
<td>0</td>
<td>N/A</td>
<td>32(^2)</td>
<td>2</td>
<td>118</td>
</tr>
</tbody>
</table>

1. Requires licensing and installation of enabling hardware.
2. Laboratory-tested to 32 trunks. Actual capacity depends on the availability of IP resources, the vocoder used, and the system configuration.

**IP and SIP Endpoints**

The Inter-Tel 5000 platform supports Inter-Tel IP phones, IP and SIP trunk gateways, and IP Single-Line Adapters (IP SLAs).

**Multi-protocol endpoints**

A multi-protocol endpoint is one that operates either in Inter-Tel Protocol (ITP) mode or in Session Initiation Protocol (SIP) mode. A SIP server is required for SIP mode operation. Multi-protocol endpoints include Models 8622, 8662, and 8690. Multi-protocol endpoints operating in ITP mode require Category A licenses. Multi-protocol endpoints operating in SIP mode require Category C licenses.

**IP and SIP Trunk Gateways**

Trunk gateways provide the signal conversion necessary for VoIP conversations to be carried across the public telephone network.

**IP Single-Line Adapters**

One IP Single-Line Adapter (IP SLA) provides the interface between a single local area network (LAN) port and one single line endpoint. Each IP SLA requires a Category C license.
Wireless Endpoints

Inter-Tel Models 8664, 8665, and 8668 are wireless IP endpoints that operate in ITP mode and require Category C licenses.

Models 8524, 8525, and 8526 are wireless digital endpoints and require Category C licenses.

Digital and Analog Endpoints

The Inter-Tel 5000 platform supports Inter-Tel digital endpoints and Single-Line Adapters (SLA), each of which supports two telephones.

Digital endpoints

The Inter-Tel 5000 platform supports the following Inter-Tel digital endpoints:

- Models 8500, 8520, and 8560
- Basic models
- Standard, Associate, Executive, Professional Display models

Single-Line Adapters

Within applicable system parameters, one SLA provides the interface between a Digital Endpoint Module (DEM-16) port and two of any of the following endpoints:

- Regular single line touchtone or dial pulse endpoints
- Recording devices
- Answering machines
- Facsimile (fax) machines
- Off-premises extensions (OPXs)

Single line endpoints

In addition being able to install single line endpoint in each of the two circuits built into the Inter-Tel 5000 Base Server, one single line endpoint can be installed with each IP SLA, and two single line endpoints can be installed with each SLA. Single line endpoints and SLAs do not require licenses, but IP SLAs require a Category C license.

Softphones

The Inter-Tel 5000 platform supports two softphone applications, which are described in the following paragraphs.

Model 8601 SoftPhone for Pocket PC

The Model 8601 is a SIP softphone application that runs on selected Pocket PC 2002, or later, Personal Digital Assistant (PDA) platforms. The Inter-Tel 5000 IPRA supports the Model 8601 but does not support legacy Inter-Tel IP SoftPhone applications. The Model 8601 SoftPhone for Pocket PC requires a Category C license.

Model 8602 IP Softphone

The Model 8602 is a softphone application that enables VoIP telephone calls from laptop or desktop computers. The Model 8602 connects to the Inter-Tel 5000 platform across the Internet and requires a Category B license.
Attendant Options

The Inter-Tel 5000 platform provides three options for monitoring, answering, transferring, or routing incoming calls.

Model 8614 Mini-DSS

The Model 8614 Mini-Digital Station Select (Mini-DSS) can be attached to a Model 8660 IP endpoint or to digital endpoint models, etc. to provide 16 viewable stations.

DSS/BLF Units

Digital Station Select/Busy Lamp Field (DSS/BLF) can be attached to several different digital endpoints to provide visibility of 50 or 60 directly accessible stations.

Automated Attendant

The Automated Attendant is an automated answering service feature. For detailed information, see “Automated Attendant” on page 265.

IP Resources

This guide refers frequently to the term “IP resources,” which has an important meaning in computer telephony (CT) jargon. In common English usage, the word “resources” refers to a supply that can be drawn on when needed. Computing resources constitute the needed supply of processing power to run CT applications, and different applications require different amounts of resources. The supply of computing resources that is not actually in use is available to run applications.

An IP resource describes the amount of computing capability in an Inter-Tel 5000 system that call processing software and hardware need to work together. Resources are used to describe an interface of a physical device used for call processing, such as an endpoint, as seen by a software application. Examples of such interfaces include a fax processing resource, a voice recognition resource, or a text-to-speech resource.

In other words, an IP resource is a quantity of processing power that digital signal processors (DSPs) on the system use for call processing functions. In terms of processing power consumed, the “cost” of a DSP function varies, depending on whether the DSP resides on the Processor Module (PM-1) or on the Processor Expansion Card (PEC-1). IP resources are used for DSP processing of speech vocoder and Caller ID transmitter functions. The Inter-Tel CS-5200 contains only the PM-1 and its associated DSP and IP resources. The Inter-Tel CS-5400, or CS-5600 Base Server, also contains the PEC-1 and its associated DSP and IP resources. The Processing Server (PS-1) unit of the Inter-Tel CS-5600 provides neither DSPs nor IP resources for system call processing.

Internet Protocol Resource Application

The Internet Protocol Resource Application (IPRA) supplies IP functionality for the system and communicates to the IP endpoints, IP trunk gateways, and other IP endpoints across 100Base-T Ethernet.
Inter-Tel 5000 IP Resource Usage

On the Inter-Tel 5000 platform, IP resources are used to provide the following call processing capabilities:

- IP endpoints
- IP network connections
- Faxing between Inter-Tel 5000 nodes

The correct allocation of IP resources is an important factor in maintaining a system that functions without calls being blocked. Field technicians can promote nonblocking system performance by making sure that the allocation of system IP resources is appropriate for the customer’s needs. System administrators having questions or concerns about the allocation of system IP resources should contact their Inter-Tel equipment and service provider.

Network Capability

Two or more Inter-Tel advanced communication platforms, Inter-Tel 5000 and/or Inter-Tel Axxess, can be connected with T1/E1/PRI network spans to form a network. To the user, the network appears as one integrated system. With few exceptions, the user can perform all of the functions across the network that can be performed within a single system or node.

In an Inter-Tel private network environment, IP endpoints can be programmed to traverse, or negotiate, Network Address Translation (NAT) tables, which may be used as a firewall. The Inter-Tel 5000 may be placed inside a NATed firewall by utilizing a static NAT assignment that allows IP endpoints to operate inside or outside the NAT/firewall. For more information about programming IP endpoints to traverse a NAT/firewall, contact your Inter-Tel service provider.

The Inter-Tel CS-5200, CS-5400, or CS-5600 can operate as a stand-alone system or as a node in a network of up to 99¹ Inter-Tel 5000 or Inter-Tel Axxess nodes. For network connectivity, Inter-Tel Axxess Converged Communications Platforms require software v7.0 or later for T1 point-to-point protocol (PPP) networking. IP networking requires v8.231 or later system software, IP Resources Card (IPRC) firmware v9.0.0 or later, and applicable licensing.

¹ The 99 node capacity requires Inter-Tel 5000 version 2.1 or Axxess version 10.0.
Network Capacities

Two or more Inter-Tel Advanced Communications Platforms (Inter-Tel 5000 or Axxess) can be connected to form a network that provides a seamless interface between the systems. To the user, the network appears as one integrated system. With few exceptions, the user can perform all of the functions across the network that can be performed within a single system.

In this guide, all references to a “network” mean two or more connected systems. Each system in a network is called a node.

The maximum capacities for networking appear in Table 44.

Table 44. System Network Capacities

<table>
<thead>
<tr>
<th>Network Node or DEVICE</th>
<th>CAPACITY</th>
</tr>
</thead>
<tbody>
<tr>
<td>System Nodes per network</td>
<td>99</td>
</tr>
<tr>
<td>Local devices per node</td>
<td>304*</td>
</tr>
<tr>
<td>Off-node devices per node</td>
<td>10,000</td>
</tr>
<tr>
<td>Devices per network</td>
<td>40,000</td>
</tr>
</tbody>
</table>

* The maximum number of endpoints and trunks that can be installed is limited by the number of voice channels and/or system memory resources available.

Near-End NAT Traversal for IP Endpoints

Near-end Network Address Translation (NAT) traversal for IP endpoints is a feature that allows an Inter-Tel 5000 platform to be placed inside a NAT or firewall and still communicate with IP endpoints outside the NAT/firewall. Near-end NAT traversal capability requires a system administrator’s intervention to configure, through DB Programming, the endpoints requiring NAT traversal. Each IP endpoint is configured for either the default Native IP address or the system’s NAT IP address. This method is appropriate for IP endpoints located in a home office or remote office.

The new Inter-Tel Model 8602 IP softphone introduces an IP endpoint that will likely need to be moved from inside a NAT/firewall to outside the NAT/firewall and back again without administrator intervention.

Automatic NAT Detection

Automatic NAT Detection introduced in v1.2 provides the capability to automatically determine whether an IP endpoint is inside the firewall or outside the firewall. In DB Programming, the administrator can now designate each IP endpoint to use the Auto type, as well as either Native or NAT types. The default selection is Auto. Automatic NAT detection is performed each time an IP endpoint comes online and whenever the feature is set through DB Programming.

For the Auto NAT Detection feature to function, the system must be running Call Processing v1.2 or later.
Licensing

Table 45 summarizes IP endpoint licensing for the Inter-Tel 5000 platform.

**Table 45. IP Endpoint Licensing Capacities**

<table>
<thead>
<tr>
<th>Type of Endpoint</th>
<th>CS-5200</th>
<th>CS-5400</th>
<th>CS-5600</th>
</tr>
</thead>
<tbody>
<tr>
<td>IP Endpoints</td>
<td>75</td>
<td>175</td>
<td>250</td>
</tr>
<tr>
<td>Configured/supported</td>
<td></td>
<td></td>
<td></td>
</tr>
<tr>
<td>IP Trunks</td>
<td>75</td>
<td>175</td>
<td>250</td>
</tr>
<tr>
<td>Configured/supported concurrently</td>
<td></td>
<td></td>
<td></td>
</tr>
<tr>
<td>IP Networking</td>
<td>3</td>
<td>6</td>
<td>6</td>
</tr>
<tr>
<td>Pre-allocated IP resources</td>
<td></td>
<td></td>
<td></td>
</tr>
<tr>
<td>Basic Voice Mail (BVM)</td>
<td>0, 4, 8</td>
<td>0, 4, 8</td>
<td>0, 4, 8, 16</td>
</tr>
<tr>
<td>Added in 4-port increments</td>
<td></td>
<td></td>
<td></td>
</tr>
</tbody>
</table>

Endpoint Capacities

Table 46 compares endpoint capacities for the Inter-Tel CS-5200, CS-5400, and CS-5600 platforms when running v1.2 and v2.x.

**Table 46. Endpoint Capacities of Inter-Tel 5000 Systems Running v1.2 and v2.x**

<table>
<thead>
<tr>
<th>System</th>
<th>v1.2</th>
<th>v2.x</th>
</tr>
</thead>
<tbody>
<tr>
<td></td>
<td>IP</td>
<td>Digital¹</td>
</tr>
<tr>
<td>Inter-Tel</td>
<td></td>
<td></td>
</tr>
<tr>
<td>CS-5200</td>
<td>28⁴</td>
<td>96</td>
</tr>
<tr>
<td>Inter-Tel</td>
<td></td>
<td></td>
</tr>
<tr>
<td>CS-5400</td>
<td>116⁴</td>
<td>96</td>
</tr>
<tr>
<td>Inter-Tel</td>
<td></td>
<td></td>
</tr>
<tr>
<td>CS-5600</td>
<td>N/A</td>
<td>110–250</td>
</tr>
</tbody>
</table>

1. Assumes two DEI units, each equipped with three DEM-16 modules.
2. Includes single line endpoints installed on ports built into the Inter-Tel 5000 Base Server but none that may be attached to IP-SLA units. Maximum 54 endpoints requires additional equipment.
3. Depending on the vocoder selected, needed capacity may require oversubscribing.
4. Total number of IP resources available without allocation for IP Networking.

Model 8602 IP Softphone

One license is required for each Inter-Tel Model 8602 softphone. Each concurrent Model 8602 requires a license.

An Enable IP Endpoint license may be required for some configurations that enable digital capability. For details, contact your authorized Inter-Tel service provider.
BVM Storage Capacities

BVM messages are stored on the compact flash-type Inter-Tel memory card that also stores system applications and the user database.

The amount of message storage, which is allocated for each user as a quota of the total storage capacity, is programmable through DB Programming. A warning flag can be set for each mailbox to alert the user when a specified percentage of the allocated quota has been reached.

Table 47 shows the amount of message storage available on the three sizes of Inter-Tel memory card before additional sets of prompts are applied. It also shows the hard disk storage capacity of BVM on PS-1 for the Inter-Tel CS-5600 platform.

When all of the IP resources available for BVM are busy, the next caller is camped-on to wait for a resource to become available. Upgrading to an expanded, external voice mail system is recommended when an unacceptable number of callers are being camped-on before they can leave a voice message.

Table 47. Basic Voice Mail Storage Capacities on Inter-Tel 5000 Systems

<table>
<thead>
<tr>
<th>Inter-Tel Memory Card w/ American English Prompts</th>
<th>BVM on PS-1 Hard Disk</th>
</tr>
</thead>
<tbody>
<tr>
<td>256 MB</td>
<td>512 MB</td>
</tr>
<tr>
<td>Inter-Tel CS-5200 or CS-5400</td>
<td>1800 min (30.00 hr)</td>
</tr>
<tr>
<td>Inter-Tel CS-5600</td>
<td>N/A</td>
</tr>
</tbody>
</table>

The BVM Forward to E-Mail feature requires significant memory for converting voice mail messages to the image file format. Inter-Tel recommends installing a 512 MB memory card for Inter-Tel CS-5200 and CS-5400 systems that utilize the feature.

In addition to the standard American English prompts, each set of language prompts loaded on a system reduces the amount of storage available by approximately 1.25 hr (75 min). For the most part, the reduction of memory capacity due to adding language prompts impacts Inter-Tel CS-5200 and CS-5400 systems. The CS-5600 relies on hard disk storage rather than the system memory card, which frees up memory for other system tasks. Inter-Tel CS-5x00 systems also support British English, Spanish, and Japanese language prompts.
### Endpoint Model-Specific Licensing

With Inter-Tel 5000 v1.2 and later, all IP endpoints are licensed in accordance with three licensing categories. Table 48 summarizes the categories and the endpoints requiring a license in each category.

**Table 48. Endpoint License Categories**

<table>
<thead>
<tr>
<th>License Category</th>
<th>Endpoints Covered</th>
</tr>
</thead>
<tbody>
<tr>
<td><strong>Category A</strong></td>
<td>Multi-protocol endpoints:</td>
</tr>
<tr>
<td>Previously licensed as Advanced IP Endpoints</td>
<td></td>
</tr>
<tr>
<td></td>
<td>• Model 8600, without display, in ITP mode</td>
</tr>
<tr>
<td></td>
<td>• Model 8622, 2-line display (and discontinued Model 8620), in ITP mode</td>
</tr>
<tr>
<td></td>
<td>• Model 8662, 6-line display, in ITP mode</td>
</tr>
<tr>
<td></td>
<td>• Model 8690 in ITP mode</td>
</tr>
<tr>
<td>Non-Inter-Tel endpoint:</td>
<td></td>
</tr>
<tr>
<td></td>
<td>• OAISYS Net Phone, which emulates an Inter-Tel Model 8662 in ITP mode</td>
</tr>
<tr>
<td><strong>Category B</strong></td>
<td>Model 8602 IP softphone</td>
</tr>
<tr>
<td><strong>Category C</strong></td>
<td>Multi-protocol endpoints:</td>
</tr>
<tr>
<td></td>
<td>• Model 8600, without display, in SIP mode</td>
</tr>
<tr>
<td></td>
<td>• Model 8622, 2-line display (and discontinued Model 8620), in SIP mode</td>
</tr>
<tr>
<td></td>
<td>• Model 8662, 6-line display, in SIP mode</td>
</tr>
<tr>
<td></td>
<td>• Model 8690 in SIP mode</td>
</tr>
<tr>
<td>Endpoints that operate in SIP mode also require SIP Server licenses.</td>
<td></td>
</tr>
<tr>
<td>Endpoints that did not require an individual license in previous versions:</td>
<td></td>
</tr>
<tr>
<td></td>
<td>• Models 8524, 8525, and 8526 wireless digital endpoints</td>
</tr>
<tr>
<td></td>
<td>• Models 8664, 8665, and 8668 wireless IP endpoints</td>
</tr>
<tr>
<td></td>
<td>• Model 8660, 6-line display, ITP mode only</td>
</tr>
<tr>
<td></td>
<td>• IP PhonePlus</td>
</tr>
<tr>
<td></td>
<td>• IP Single-Line Adapter (IP SLA)</td>
</tr>
<tr>
<td>SIP endpoints:</td>
<td></td>
</tr>
<tr>
<td></td>
<td>• Model 8601 SoftPhone for Pocket PC (Operates in SIP mode only)</td>
</tr>
<tr>
<td>Other:</td>
<td></td>
</tr>
<tr>
<td></td>
<td>• Enterprise® Conferencing port</td>
</tr>
<tr>
<td></td>
<td>• Cisco® 7960 SIP Phone</td>
</tr>
</tbody>
</table>
## System Alarms

### Table 49. System Alarms

<table>
<thead>
<tr>
<th>Alarm Text</th>
<th>Description and Action Required</th>
</tr>
</thead>
<tbody>
<tr>
<td>SYS ALARM #10</td>
<td>An endpoint has remained off-hook and inactive past the expiration of the Inactivity Alarm timer. The display and the SMDR indicate which endpoint is off-hook. The lamps in the button for that station on Mini-DSS and DSS units and on the Speed Dial buttons of other endpoints flutter continuously. Locate the endpoint and replace the handset in the cradle. Calls being transmitted over the secondary voice path are not affected or interrupted by an off-hook alarm condition.</td>
</tr>
<tr>
<td>xNNNN OFF HOOK</td>
<td></td>
</tr>
<tr>
<td>SYS ALARM #11</td>
<td>A user has dialed the Emergency Call feature code. The system has seized a trunk and dialed the emergency phone number that will bring responders to the location where the system Inter-Tel 5000 Base Server is installed. The default Emergency Call code for U.S. systems is 911, and the default code for European systems is 999. Emergency dial sequences, such as 112 for some European localities, can be programmed for the location where the system Inter-Tel 5000 Base Server is installed. However, if an appropriate emergency phone number is dialed from a remote location to ensure response to that location, no alarm will appear at the administrator(s). No action is required, just clear the alarm. To facilitate emergency services reaching the party in need, determine who made the emergency call and where help is needed.</td>
</tr>
<tr>
<td>xNNNN EMERGENCY</td>
<td></td>
</tr>
<tr>
<td>SYS ALARM #12</td>
<td>The printer is not functioning properly. Check the cable and the power cord connections and that the printer is supplied with paper and toner.</td>
</tr>
<tr>
<td>CHECK PRINTER</td>
<td></td>
</tr>
<tr>
<td>SYS ALARM #13</td>
<td>The printer is not functioning properly. Check the cable and the power cord connections and that the printer is supplied with paper and toner.</td>
</tr>
<tr>
<td>CHECK PRINTER</td>
<td></td>
</tr>
<tr>
<td>SYS ALARM #14</td>
<td>The system real-time clock has not been initialized or has been corrupted. Set the system time and date as described on page 28.</td>
</tr>
<tr>
<td>SET DATE/TIME</td>
<td></td>
</tr>
<tr>
<td>SYS MGR INV CERT</td>
<td>A corrupt certificate was uploaded or there was an error in writing the file. Contact service personnel.</td>
</tr>
<tr>
<td>SYS ALARM #18</td>
<td></td>
</tr>
<tr>
<td>SYS MGR LINK ERR</td>
<td>The certificate is not yet valid or the system is not connected to the appropriate System Manager server. Contact service personnel.</td>
</tr>
<tr>
<td>&lt;error string&gt;</td>
<td></td>
</tr>
<tr>
<td>SYS ALARM #15</td>
<td>The real-time clock chip is not operating. Contact service personnel.</td>
</tr>
<tr>
<td>CALL TECHNICIAN</td>
<td></td>
</tr>
<tr>
<td>SYS ALARM #16</td>
<td>The system was unable to perform a scheduled delayed major reset because the system was not idle. If this alarm recurs, contact service personnel.</td>
</tr>
<tr>
<td>NO DELAYED MAJOR</td>
<td></td>
</tr>
<tr>
<td>SYS MGR LINK ERR</td>
<td>The certificate is not yet valid or the system is not connected to the appropriate System Manager server. Contact service personnel.</td>
</tr>
<tr>
<td>SYS MGR INV CERT</td>
<td></td>
</tr>
</tbody>
</table>
### Table 49. System Alarms (Continued)

<table>
<thead>
<tr>
<th>Alarm Text</th>
<th>Description and Action Required (Continued)</th>
</tr>
</thead>
<tbody>
<tr>
<td>SYS ALARM #20</td>
<td>The printer is not functioning properly. Check that the cable and the power cord are connected, and that it has paper and toner.</td>
</tr>
<tr>
<td>CHECK PRINTER</td>
<td></td>
</tr>
<tr>
<td>SYS ALARM #21</td>
<td>The external voice mail disk space is 80% of full. If it reaches 100%, Voice Mail cannot accept any messages until disk space is made available.</td>
</tr>
<tr>
<td>V-MAIL 80% FULL</td>
<td>Contact service personnel.</td>
</tr>
<tr>
<td>SYS ALARM #22</td>
<td>100% of the external voice mail disk space is in use. Voice mail cannot accept any messages until disk space is made available.</td>
</tr>
<tr>
<td>V-MAIL 100% FULL</td>
<td>Contact service personnel.</td>
</tr>
<tr>
<td>SYS ALARM #23</td>
<td>The SMDR disk space on the external voice processing system is 80% of full. If it reaches 100% before the SMDR information is cleared, no further SMDR recording will be possible.</td>
</tr>
<tr>
<td>SMDR 80% FULL</td>
<td>Contact service personnel.</td>
</tr>
<tr>
<td>SYS ALARM #24</td>
<td>100% of the SMDR disk space on the external voice processing system is in use. No further SMDR recording is possible.</td>
</tr>
<tr>
<td>SMDR 100% FULL</td>
<td>Contact service personnel.</td>
</tr>
<tr>
<td>SYS ALARM #32</td>
<td>The IP Network does not have enough bandwidth to support the IP call that is currently connected to the extension. If this alarm is generated frequently, it could indicate a network problem that requires the attention of your network administrator.</td>
</tr>
<tr>
<td>INSUF BAND</td>
<td>Notify your network administrator. The affected device is not behind a firewall or NAT.</td>
</tr>
</tbody>
</table>
Administrator Programming Planning Sheets

The following program planning sheets are designed to help you plan your site’s system and voice mail programming.

Custom Audiotex Recordings

If you customize Audiotex recordings, use the table below to make note of the text. For further instructions, see “Creating Audiotex Recordings” on page 71. Depending on which voice processing system your site uses, how your system is programmed, and how the recordings are used, an Automatic Speech Recognition (ASR) version, as well as a non-ASR version, may be needed.

<table>
<thead>
<tr>
<th>Recording #</th>
<th>Text</th>
<th>Where Used (Ext. #)</th>
</tr>
</thead>
<tbody>
<tr>
<td>Example</td>
<td>Thank you for calling XYZ Company. All of our representatives are busy right now. Please hold.</td>
<td>Ext.2011 (Hunt Group Overflow Station)</td>
</tr>
</tbody>
</table>
Endpoint Programming

Administrator Database (DB) Programming Password: ___________________________

Endpoint Information

For each station to be programmed, record the following information.

<table>
<thead>
<tr>
<th>Endpoint Extension #:</th>
<th>Administrator endpoint?</th>
<th>ATT: Yes or No</th>
</tr>
</thead>
<tbody>
<tr>
<td></td>
<td>Attendant endpoint?</td>
<td></td>
</tr>
<tr>
<td></td>
<td>House Phone?</td>
<td>Yes or No</td>
</tr>
<tr>
<td></td>
<td>This endpoint's attendant:</td>
<td></td>
</tr>
<tr>
<td></td>
<td>Username:</td>
<td></td>
</tr>
<tr>
<td></td>
<td>Toll restriction classes of service:</td>
<td></td>
</tr>
</tbody>
</table>

Do-Not-Disturb Messages

<table>
<thead>
<tr>
<th>Default DND Message</th>
<th>Primary Language DND Message</th>
<th>Secondary Language DND Message</th>
</tr>
</thead>
<tbody>
<tr>
<td>MESSAGE 01 (DO-NOT-DISTURB)</td>
<td></td>
<td></td>
</tr>
<tr>
<td>MESSAGE 02 (LEAVE A MESSAGE)</td>
<td></td>
<td></td>
</tr>
<tr>
<td>MESSAGE 03 (IN MEETING UNTIL)</td>
<td></td>
<td></td>
</tr>
<tr>
<td>MESSAGE 04 (IN MEETING)</td>
<td></td>
<td></td>
</tr>
<tr>
<td>MESSAGE 05 (ON VACATION/ HOLIDAY 'TIL)</td>
<td></td>
<td></td>
</tr>
<tr>
<td>MESSAGE 06 (ON VACATION HOLIDAY)</td>
<td></td>
<td></td>
</tr>
<tr>
<td>MESSAGE 07 (CALL ME AT)</td>
<td></td>
<td></td>
</tr>
<tr>
<td>MESSAGE 08 (AT THE DOCTOR)</td>
<td></td>
<td></td>
</tr>
<tr>
<td>MESSAGE 09 (ON A TRIP)</td>
<td></td>
<td></td>
</tr>
<tr>
<td>MESSAGE 10 (ON BREAK)</td>
<td></td>
<td></td>
</tr>
<tr>
<td>MESSAGE 11 (OUT OF TOWN 'TIL)</td>
<td></td>
<td></td>
</tr>
<tr>
<td>MESSAGE 12 (OUT OF OFFICE)</td>
<td></td>
<td></td>
</tr>
<tr>
<td>MESSAGE 13 (OUT UNTIL)</td>
<td></td>
<td></td>
</tr>
<tr>
<td>MESSAGE 14 (WITH A CLIENT)</td>
<td></td>
<td></td>
</tr>
<tr>
<td>MESSAGE 15 (WITH A GUEST)</td>
<td></td>
<td></td>
</tr>
<tr>
<td>MESSAGE 16 (UNAVAILABLE)</td>
<td></td>
<td></td>
</tr>
<tr>
<td>MESSAGE 17 (IN CONFERENCE)</td>
<td></td>
<td></td>
</tr>
<tr>
<td>MESSAGE 18 (AWAY FROM DESK)</td>
<td></td>
<td></td>
</tr>
<tr>
<td>MESSAGE 19 (GONE HOME)</td>
<td></td>
<td></td>
</tr>
<tr>
<td>MESSAGE 20 (OUT TO LUNCH)</td>
<td></td>
<td></td>
</tr>
</tbody>
</table>
## Reminder Messages

<table>
<thead>
<tr>
<th>Default Reminder Message</th>
<th>Primary Language Reminder Message</th>
<th>Secondary Language Reminder Message</th>
</tr>
</thead>
<tbody>
<tr>
<td>MESSAGE 01 (MEETING):</td>
<td></td>
<td></td>
</tr>
<tr>
<td>MESSAGE 02 (STAFF MEETING):</td>
<td></td>
<td></td>
</tr>
<tr>
<td>MESSAGE 03 (SALES MEETING):</td>
<td></td>
<td></td>
</tr>
<tr>
<td>MESSAGE 04 (CANCEL MEETING):</td>
<td></td>
<td></td>
</tr>
<tr>
<td>MESSAGE 05 (APPOINTMENT):</td>
<td></td>
<td></td>
</tr>
<tr>
<td>MESSAGE 06 (PLACE CALL):</td>
<td></td>
<td></td>
</tr>
<tr>
<td>MESSAGE 07 (CALL CLIENT):</td>
<td></td>
<td></td>
</tr>
<tr>
<td>MESSAGE 08 (CALL CUSTOMER):</td>
<td></td>
<td></td>
</tr>
<tr>
<td>MESSAGE 09 (CALL HOME):</td>
<td></td>
<td></td>
</tr>
<tr>
<td>MESSAGE 10 (CALL CORPORATE):</td>
<td></td>
<td></td>
</tr>
<tr>
<td>MESSAGE 11 (CALL ENGINEERING):</td>
<td></td>
<td></td>
</tr>
<tr>
<td>MESSAGE 12 (CALL MARKETING):</td>
<td></td>
<td></td>
</tr>
<tr>
<td>MESSAGE 13 (CALL ACCOUNTING):</td>
<td></td>
<td></td>
</tr>
<tr>
<td>MESSAGE 14 (CANCEL DND):</td>
<td></td>
<td></td>
</tr>
<tr>
<td>MESSAGE 15 (CANCEL CALL FWD):</td>
<td></td>
<td></td>
</tr>
<tr>
<td>MESSAGE 16 (TAKE MEDICATION):</td>
<td></td>
<td></td>
</tr>
<tr>
<td>MESSAGE 17 (MAKE RESERVATION):</td>
<td></td>
<td></td>
</tr>
<tr>
<td>MESSAGE 18 (REVIEW SCHEDULE):</td>
<td></td>
<td></td>
</tr>
<tr>
<td>MESSAGE 19 (LUNCH):</td>
<td></td>
<td></td>
</tr>
<tr>
<td>MESSAGE 20 (REMINDER):</td>
<td></td>
<td></td>
</tr>
</tbody>
</table>
### Changing Station Extension Numbers

<table>
<thead>
<tr>
<th>Old Extension Number</th>
<th>New Extension Number</th>
</tr>
</thead>
<tbody>
<tr>
<td></td>
<td></td>
</tr>
<tr>
<td></td>
<td></td>
</tr>
<tr>
<td></td>
<td></td>
</tr>
<tr>
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<td></td>
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<tr>
<td></td>
<td></td>
</tr>
<tr>
<td></td>
<td></td>
</tr>
</tbody>
</table>

### Swapping Stations

<table>
<thead>
<tr>
<th>Extension Number</th>
<th>Swapped with Extension</th>
</tr>
</thead>
<tbody>
<tr>
<td></td>
<td></td>
</tr>
<tr>
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</tr>
</tbody>
</table>
## Individual Trunk Programming

Record the following information for each trunk to be programmed.

<table>
<thead>
<tr>
<th>Information Required</th>
<th>Responses</th>
</tr>
</thead>
<tbody>
<tr>
<td>Answer Supervision Type:</td>
<td>Polarity Reversal or Valid Call or Valid Call with Polarity Reversal</td>
</tr>
<tr>
<td>Caller ID?</td>
<td>Yes or No</td>
</tr>
<tr>
<td>Hybrid Balance:</td>
<td>Normal or Short</td>
</tr>
<tr>
<td>Signaling:</td>
<td>DTMF or Dial Pulse</td>
</tr>
<tr>
<td>Trunk Group Number:</td>
<td></td>
</tr>
<tr>
<td>Username:</td>
<td></td>
</tr>
</tbody>
</table>
Trunk Group Programming

Record the following information for each trunk group to be programmed.

<table>
<thead>
<tr>
<th>Information Required</th>
<th>Programming Plan</th>
</tr>
</thead>
<tbody>
<tr>
<td>Endpoints with Day Answer Access:</td>
<td></td>
</tr>
<tr>
<td>Endpoints with Night Answer Access:</td>
<td></td>
</tr>
<tr>
<td>Endpoints with Day Ring In:</td>
<td></td>
</tr>
<tr>
<td>Endpoints with Night Ring In:</td>
<td></td>
</tr>
<tr>
<td>Toll Restriction Classes of Service:</td>
<td></td>
</tr>
<tr>
<td>Trunks in Trunk Groups:</td>
<td></td>
</tr>
</tbody>
</table>

**NOTICE**

IP and SIP endpoints at remote sites require gateway trunks, gateways, and appropriate programming to reach their local emergency responders.
Default Feature Codes

Each of the station features is assigned a feature code. Using the station’s dialpad, these codes are entered to select trunks, process calls, and use special features. The following tables show default feature code values:

- Table 50 on this page.
- Table 51 on page 316.
- Table 52 on page 322.
- Table 53 on page 324.

Trunk Access Codes

These feature codes are used to select trunks when placing outgoing calls.

Table 50. Trunk Access Codes

<table>
<thead>
<tr>
<th>Feature Name</th>
<th>Code U.S. [Europe]</th>
<th>Definition</th>
</tr>
</thead>
<tbody>
<tr>
<td>Automatic Route Selection (ARS)</td>
<td>92000</td>
<td>Allows the system to select the route wanted for placing a call, as programmed in the database.</td>
</tr>
<tr>
<td>Trunk Group Access 1–208</td>
<td>92001–92208</td>
<td>Selects an available trunk from a programmed group of trunks for placing an outside call.</td>
</tr>
<tr>
<td>Emergency Call</td>
<td>911 [999 or as applicable]</td>
<td>Entering this feature code selects an outgoing trunk and automatically dials the programmed Emergency Call number, which is routed by default out Trunk Group 1.</td>
</tr>
<tr>
<td>Outgoing Call</td>
<td>8</td>
<td>Selects an outgoing trunk according to the programmed outgoing access mode for that endpoint. Inter-Tel endpoints use the OUT-GOING button or a Call button.</td>
</tr>
</tbody>
</table>
Endpoint Feature Codes

Table 51 summarizes endpoint feature codes. When a default feature button can be used in place of the feature code, it is indicated at the end of the explanation. For a list of the default endpoint feature buttons, see Table 15 on page 91.

**Table 51. Endpoint Feature Codes**

<table>
<thead>
<tr>
<th>Feature Name</th>
<th>Code</th>
<th>Definition</th>
</tr>
</thead>
<tbody>
<tr>
<td>Account Code – All Calls Following</td>
<td>391</td>
<td>Allows the endpoint user to enter a forced or optional account code that will apply to all calls following the entry of this feature code and will appear in the SMDR. To disable the All Calls Following feature, the feature code is entered again without an account code.</td>
</tr>
<tr>
<td>Account Code – Optional</td>
<td>390</td>
<td>Allows the endpoint user to enter an optional account code for SMDR reports during an outside call.</td>
</tr>
<tr>
<td>ACD Agent Login</td>
<td>326</td>
<td>These feature codes allow an ACD hunt group member (agent) to log into and out of the ACD hunt group(s). The agent will only receive calls through the ACD hunt group(s) while logged in.</td>
</tr>
<tr>
<td>ACD Agent Logout</td>
<td>327</td>
<td></td>
</tr>
<tr>
<td>ACD Agent Login/Logout Toggle</td>
<td>328</td>
<td></td>
</tr>
<tr>
<td>ACD Agent Wrap-Up Terminate</td>
<td>329</td>
<td>When an ACD agent completes a call, no other ACD hunt group call will ring at the endpoint until the ACD Wrap-Up Timer expires or the agent enters this feature code to terminate the wrap-up session.</td>
</tr>
<tr>
<td>Agent Help</td>
<td>375</td>
<td>The Agent Help feature allows an endpoint user to request help from a designated “Agent Help Extension” during a two- or three-party call.</td>
</tr>
<tr>
<td>Agent Help Reject</td>
<td>376</td>
<td>When a request-for-help call rings, the Agent Help Extension can choose to join the call or enter this feature code to reject the request.</td>
</tr>
<tr>
<td>Answer (Ringing Call)</td>
<td>351</td>
<td>Answers the call that has been ringing or holding the longest at that endpoint. See page 139 for the priority list. (Inter-Tel endpoints use the ANSWER button.)</td>
</tr>
<tr>
<td>Audio Diagnostics</td>
<td>320</td>
<td>When initiated, users are prompted to answer questions about the audio problems by pressing specific keypad buttons.</td>
</tr>
<tr>
<td>Automatic CO Access On/Off</td>
<td>360</td>
<td><em>(Not used on single line endpoints)</em> Allows the endpoint user to determine how ringing outside calls will be answered: simply by lifting the handset or pressing the Speaker button (automatic answer), or by lifting the handset or pressing the Speaker button and pressing a Call button, individual trunk button or the ANSWER button.</td>
</tr>
</tbody>
</table>
### Table 51. Endpoint Feature Codes (Continued)

<table>
<thead>
<tr>
<th>Feature Name</th>
<th>Code</th>
<th>Definition</th>
</tr>
</thead>
<tbody>
<tr>
<td>Automatic IC Access On/Off</td>
<td>361</td>
<td><em>(Not used on single line endpoints)</em> Allows the endpoint user to determine how ringing intercom calls will be answered: simply by lifting the handset (automatic answer), or by lifting the handset and pressing the IC button (or a Call button, if there is no IC button).</td>
</tr>
<tr>
<td>Automatic Trunk Answer</td>
<td>350</td>
<td>Using this feature code, endpoint users with allowed answer can pick up trunks that are ringing into the system, but that are not actually ringing at their endpoints. This feature does not pick up transferred calls or recalls that are ringing at the endpoint.</td>
</tr>
<tr>
<td>Background Music On/Off</td>
<td>313</td>
<td><em>(Not used on single line endpoints)</em> Turns on and off background music heard through the endpoint speaker.</td>
</tr>
<tr>
<td>Barge-In</td>
<td>386</td>
<td>Allows the supervisor to barge-in on a call to help the hunt group member/agent.</td>
</tr>
<tr>
<td>Call Forward All Calls</td>
<td>355</td>
<td>Immediately forwards all calls to another endpoint or to an outside endpoint number. <em>(Inter-Tel endpoints use the FWD button.)</em></td>
</tr>
<tr>
<td>Call Forward If Busy</td>
<td>357</td>
<td>Immediately forwards all calls to another endpoint or to an outside endpoint number when the endpoint is in use.</td>
</tr>
<tr>
<td>Call Forward If No Answer</td>
<td>356</td>
<td>Forwards all calls to another endpoint or to an outside endpoint number if not answered within a predetermined time.</td>
</tr>
<tr>
<td>Call Forward If No Answer/Busy</td>
<td>358</td>
<td>Forwards all calls to another endpoint or to an outside endpoint number if not answered within a predetermined amount of time, or immediately if the endpoint is in use.</td>
</tr>
<tr>
<td>Call Logging</td>
<td>333</td>
<td>Allows users of display endpoints to view missed, received, and dialed calls.</td>
</tr>
<tr>
<td>Change Language</td>
<td>301</td>
<td>An endpoint user can change the assigned language for the endpoint by entering the Change Language feature code while the endpoint is idle.</td>
</tr>
<tr>
<td>CO Hookflash</td>
<td>330</td>
<td>Sends a timed hookflash over the trunk while on an outside call (includes conference calls).</td>
</tr>
<tr>
<td>Conference</td>
<td>5</td>
<td>Connects from three to four parties in a conference. A conference consists of any combination of inside and outside parties. <em>(Inter-Tel endpoints use the CNF button.)</em></td>
</tr>
<tr>
<td>Data</td>
<td>340</td>
<td>Allows operation of a data device attached to a digital endpoint. Requires a modem-equipped data device. Supported on Inter-Tel 5000 systems equipped with appropriate digital interface equipment.</td>
</tr>
</tbody>
</table>
Table 51. Endpoint Feature Codes (Continued)

<table>
<thead>
<tr>
<th>Feature Name</th>
<th>Code</th>
<th>Definition</th>
</tr>
</thead>
<tbody>
<tr>
<td>Default Endpoint</td>
<td>394</td>
<td>This single feature code cancels account codes for all calls following, Do-Not-Disturb, manual call forwarding, background music, ring intercom always, and queue requests; restores handsfree mode, pages, hunt group calls, and system forwarding; and returns endpoint volumes to default values.</td>
</tr>
<tr>
<td>Directories</td>
<td>307</td>
<td><em>(Display endpoints Only)</em> Allows display endpoint users to search for extension numbers or System Speed Dial numbers. The number can then be dialed, if appropriate.</td>
</tr>
<tr>
<td>Display Outside Party Name On/Off</td>
<td>379</td>
<td><em>(Display endpoints Only)</em> When the endpoint user enters this feature code, while connected to a CO call that has outside party name information, the display will toggle between the caller’s name and number. If there is no outside party name or the Expanded CO Call Information On Displays flag is disabled, the user will hear a burst of reorder tone and see the CANNOT ACCESS FEATURE display. If the Display Outside Party Name On/Off feature code is programmed in a user programmable button with a lamp, the lamp will be lit when the outside party name is enabled and off when the outside party number is enabled.</td>
</tr>
<tr>
<td>Display Time/Date (ITP)</td>
<td>300</td>
<td><em>(Display endpoints Only)</em> Temporarily displays the system date and time, username, and extension number during a call or when other displays are shown. Feature code 300 displays the IP Address of an endpoint if it is in SIP mode.</td>
</tr>
<tr>
<td>Do-Not-Disturb</td>
<td>370</td>
<td></td>
</tr>
<tr>
<td>Do-Not-Disturb Cancel</td>
<td>371</td>
<td></td>
</tr>
<tr>
<td>Do-Not-Disturb On/Off</td>
<td>372</td>
<td>The Do-Not-Disturb feature code halts all intercom calls, transferred calls, and pages to the endpoint. The Cancel code returns the endpoint to normal operation. The on/off code can be used to turn Do-Not-Disturb on or off. <em>(Inter-Tel endpoints use the DND button).</em></td>
</tr>
<tr>
<td>Do-Not-Disturb Override</td>
<td>373</td>
<td><em>(Not used on single line endpoints)</em> If enabled in the database, allows the endpoint user to break through another endpoint’s Do-Not-Disturb mode when placing an intercom call.</td>
</tr>
<tr>
<td>Enhanced Speakerphone Enable</td>
<td>310</td>
<td><em>(Digital endpoints only)</em> When entered at a digital endpoint, this feature code enables the enhanced speakerphone as described in “Display and Nondisplay Digital Endpoints” on page 98. Digital endpoints can also use the Special button + Speaker buttons.</td>
</tr>
<tr>
<td>Feature Key Default</td>
<td>395</td>
<td><em>(Not used on single line endpoints)</em> Endpoints have user-programmable feature buttons that can be set to enter feature codes. This code returns the user-programmable buttons to the database default values.</td>
</tr>
</tbody>
</table>
### Table 51. Endpoint Feature Codes (Continued)

<table>
<thead>
<tr>
<th>Feature Name</th>
<th>Code</th>
<th>Definition</th>
</tr>
</thead>
<tbody>
<tr>
<td>Group Listen</td>
<td>312</td>
<td><em>(Not used on single line endpoints)</em> Allows a user to transmit a conversation over the endpoint speaker while in handset or headset mode.</td>
</tr>
<tr>
<td>Handsfree On/Off</td>
<td>319</td>
<td><em>(Not used on single line endpoints)</em> Disables/enables the endpoint’s handsfree intercom answering. Incoming intercom calls ring as private calls if handsfree answering is disabled.</td>
</tr>
<tr>
<td>Headset Enable</td>
<td>315</td>
<td><em>(Not used on single line endpoints)</em> The enable code signals the system that a headset has been connected to the endpoint. The disable code returns the endpoint to normal operation. The on/off feature code can be used to toggle the feature on or off.</td>
</tr>
<tr>
<td>Headset Disable</td>
<td>316</td>
<td></td>
</tr>
<tr>
<td>Headset On/Off</td>
<td>317</td>
<td></td>
</tr>
<tr>
<td>Hold – Individual</td>
<td>336</td>
<td>Places a call on hold so that it can be picked up directly at that endpoint or through a reverse transfer from any other endpoint. <em>(Inter-Tel endpoints use the HOLD button.)</em></td>
</tr>
<tr>
<td>Hold – System</td>
<td>335</td>
<td>Places an outside call on system hold. It can be picked up directly at any endpoint that has an individual trunk button and has allowed-answer and/or outgoing access for that trunk, or by the endpoint that placed it on hold. <em>(If used on conference or intercom calls, the system places the call on individual hold.)</em></td>
</tr>
<tr>
<td>Hunt Group Remove</td>
<td>322</td>
<td>Removes the endpoint from its assigned hunt group(s) or places it in again. Does not affect non-hunt group calls. The remove/replace feature code can be used to toggle the feature.</td>
</tr>
<tr>
<td>Hunt Group Replace</td>
<td>323</td>
<td></td>
</tr>
<tr>
<td>Hunt Group Remove/Replace</td>
<td>324</td>
<td></td>
</tr>
<tr>
<td>LCD Contrast Adjustment</td>
<td>303</td>
<td><em>(Model 8620/8622, 8660, or 8662)</em> Adjusts the LCD contrast on the display. The endpoint must be idle to use this feature.</td>
</tr>
<tr>
<td>Message</td>
<td>365</td>
<td>This feature code is used for leaving and retrieving a message waiting indication at a called endpoint or the called endpoint’s message center. Depending on how the message was left, the called endpoint user either retrieves the message from his/her message center or from the endpoint that left the message. <em>(Inter-Tel endpoints use the MSG button.)</em></td>
</tr>
<tr>
<td>Message – Cancel</td>
<td>366</td>
<td>Allows the endpoint user to cancel a message waiting indication that he or she left at another endpoint.</td>
</tr>
<tr>
<td>Message – Cancel Current</td>
<td>368</td>
<td>Cancels a message waiting indication that is waiting at the endpoint without requiring the user to respond to it. <em>(Inter-Tel endpoints can use the asterisk [</em>] button while viewing the message.)*</td>
</tr>
</tbody>
</table>
Table 51. Endpoint Feature Codes (Continued)

<table>
<thead>
<tr>
<th>Feature Name</th>
<th>Code</th>
<th>Definition</th>
</tr>
</thead>
<tbody>
<tr>
<td>Message – Silent</td>
<td>367</td>
<td>Leaves a Message Waiting indication at an endpoint without first placing an intercom call.</td>
</tr>
<tr>
<td>Mute On/Off</td>
<td>314</td>
<td>(Not used on single line endpoints) Turns the microphone on or off during a call. If muted, the endpoint user can hear the other party, but the party cannot hear the endpoint user. (Inter-Tel endpoints can also use the MUTE button.)</td>
</tr>
<tr>
<td>Page</td>
<td>7</td>
<td>When followed by a paging zone code (0–9 or 0–49), it allows announcements to be made through endpoint speakers and any external paging speakers in the page zone.</td>
</tr>
<tr>
<td>Page On/Off</td>
<td>325</td>
<td>(Not used on single line endpoints) Halts pages through the endpoint speaker or allows them to be received again.</td>
</tr>
<tr>
<td>Program Buttons</td>
<td>397</td>
<td>(Not used on single line endpoints) User-programmable feature buttons and Station Speed Dial buttons can be programmed using this feature code.</td>
</tr>
<tr>
<td>Program Endpoint Password</td>
<td>392</td>
<td>The endpoint password is used for the Remote Programming feature. The password can be changed by entering the Program Endpoint Password feature code at the endpoint or when using the Remote Programming feature.</td>
</tr>
<tr>
<td>Queue Request</td>
<td>6</td>
<td>Requests (or cancels) an automatic callback when a busy trunk or endpoint becomes available.</td>
</tr>
<tr>
<td>Record-A-Call</td>
<td>385</td>
<td>If the system is programmed with a Record-A-Call application, the endpoints can be programmed to use the Record-A-Call feature. It allows users to enter a feature code whenever they want to record an ongoing call in their designated Record-A-Call mailbox. Users can retrieve the recorded messages later, just as they would any other mailbox messages.</td>
</tr>
<tr>
<td>Redial</td>
<td>380</td>
<td>Redials the last outside phone number dialed or saved at the endpoint (up to 48 digits). Also used to save numbers at endpoints programmed for last number saved. (Inter-Tel endpoints use the REDIAL button.)</td>
</tr>
<tr>
<td>Redirect Call</td>
<td>331</td>
<td>Allows the endpoint user to route ringing outside, intercom, and camped on calls to another endpoint, hunt group, or outside number. Routing of the redirected call is still subject to trunk and toll restrictions. This feature provides these options in addition to the currently available options which allow the endpoint user to redirect calls to Voice Mail or Do-Not-Disturb. The Redirect Ringing Call feature does not require any premium features.</td>
</tr>
</tbody>
</table>
### Table 51. Endpoint Feature Codes (Continued)

<table>
<thead>
<tr>
<th>Feature Name</th>
<th>Code</th>
<th>Definition</th>
</tr>
</thead>
<tbody>
<tr>
<td>Reminder Message</td>
<td>305</td>
<td>(Not used on single line endpoints) The endpoint user can set reminder messages that signal the endpoint at specific times. Or, the user can cancel all reminder messages for the endpoint.</td>
</tr>
<tr>
<td>Reminder Message Cancel</td>
<td>306</td>
<td></td>
</tr>
<tr>
<td>Remote Programming</td>
<td>359</td>
<td>Allows a user to place an endpoint in DND mode, forward the endpoint’s calls, or change the password; either from another endpoint or through DISA.</td>
</tr>
<tr>
<td>Reverse Transfer (Call Pick-Up)</td>
<td>4</td>
<td>Picks up a call ringing or holding at an endpoint or hunt group</td>
</tr>
<tr>
<td>Review Keys</td>
<td>396</td>
<td>(Not used on single line endpoints) User-programmable feature buttons and Station Speed Dial buttons can be viewed using this feature code.</td>
</tr>
<tr>
<td>Ring Intercom Always On/Off</td>
<td>377</td>
<td>Enables/disables the feature that allows the endpoint to always place private (non-hands-free) intercom calls.</td>
</tr>
<tr>
<td>Ring Tone Selection</td>
<td>398</td>
<td>(Not used on single line endpoints) Selects the type of ringing alert tone that will be heard from the endpoint.</td>
</tr>
<tr>
<td>Routing Off</td>
<td>304</td>
<td>Enables System OAI Offering Control for third-party applications. This feature requires you to enter a password. Once you disable routing, you cannot enable it again (i.e., only the third-party application can enable routing).</td>
</tr>
<tr>
<td>Station Monitor</td>
<td>321</td>
<td>(Hunt Group supervisors only) Allows a designated hunt group supervisor to monitor a call of anyone in the associated hunt group.</td>
</tr>
<tr>
<td>Station Speed Dial Programming</td>
<td>382</td>
<td>Dials/programs one of the 10 Station Speed Dial numbers when followed by a location code (0–9). Inter-Tel endpoints use the Station Speed Dial button plus a location code for programming and dialing, or they can program Speed Dial buttons for one-button dialing.</td>
</tr>
<tr>
<td>Steal</td>
<td>387</td>
<td>Allows the supervisor to take away a call from the hunt group member/agent.</td>
</tr>
<tr>
<td>Switch Keymap</td>
<td>399</td>
<td>(Not used on single line endpoints) Allows an endpoint user to switch between standard and alternate keymaps.</td>
</tr>
<tr>
<td>System Forward Enable</td>
<td>352</td>
<td>Enables or disables the database-programmed System Forwarding feature for this endpoint. The on/off feature code can be used to toggle the feature on or off.</td>
</tr>
<tr>
<td>System Forward Disable</td>
<td>353</td>
<td></td>
</tr>
<tr>
<td>System Forward On/Off</td>
<td>354</td>
<td></td>
</tr>
<tr>
<td>System Speed Dial</td>
<td>381</td>
<td>Dials one of the 1000 System Speed Dial phone numbers when followed by a location code (000–999). Also used for reviewing System Speed Dial numbers.</td>
</tr>
</tbody>
</table>
Table 51. Endpoint Feature Codes (Continued)

<table>
<thead>
<tr>
<th>Feature Name</th>
<th>Code</th>
<th>Definition</th>
</tr>
</thead>
<tbody>
<tr>
<td>Transfer to Hold</td>
<td>346</td>
<td>Transfers a call to another endpoint and places it on individual hold so that it does not ring or send call waiting signals until it recalls.</td>
</tr>
<tr>
<td>Transfer to Ring</td>
<td>345</td>
<td>Transfers a call to another endpoint or to an outside phone number.</td>
</tr>
</tbody>
</table>

Administrator Feature Codes

Table 52 summarizes feature codes accessible only on administrator endpoints. Non-administrator endpoints produce reorder/fast-busy tones if these feature codes are keyed into them.

Table 52. Administrator Feature Codes

<table>
<thead>
<tr>
<th>Feature Name</th>
<th>Code</th>
<th>Definition</th>
</tr>
</thead>
<tbody>
<tr>
<td>Clear Network Alarm</td>
<td>9851</td>
<td>Clears network-wide alarms on every node in the network, but will not affect system alarms. The Clear Network Alarm feature code may be entered on any node in the network, but the Send Network Alarms flag must be set for the administrator to clear alarms on other nodes in the network.</td>
</tr>
<tr>
<td>Clear System Alarm</td>
<td>9850</td>
<td>Clears all local and network-wide system alarm displays on your node.</td>
</tr>
<tr>
<td>Diagnostics On/Off</td>
<td>9900</td>
<td>Enables or disables system diagnostics mode to perform diagnostics functions such as freezing/unfreezing the database, printing message logs, and seizing a device for troubleshooting purposes.</td>
</tr>
<tr>
<td>Diagnostics On/Off</td>
<td>[9100]</td>
<td>Enables or disables system diagnostics mode to perform diagnostics functions such as freezing/unfreezing the database, printing message logs, and seizing a device for troubleshooting purposes.</td>
</tr>
<tr>
<td>Enable Network Day</td>
<td>9862</td>
<td>Places any or all remote nodes into day mode (you are prompted for the node number).</td>
</tr>
<tr>
<td>Enable Network Night</td>
<td>9861</td>
<td>Places any or all remote nodes into night mode (you are prompted for the node number).</td>
</tr>
<tr>
<td>Modem Disable</td>
<td>9867</td>
<td>Disables the internal modem; used in conjunction with the assigned extension number of the modem.</td>
</tr>
<tr>
<td>Modem Enable</td>
<td>9866</td>
<td>Enables the internal modem; used in conjunction with the assigned extension number of the modem.</td>
</tr>
</tbody>
</table>
### Table 52. Administrator Feature Codes (Continued)

<table>
<thead>
<tr>
<th>Feature Name</th>
<th>Code</th>
<th>U.S. [Europe]</th>
<th>Definition</th>
</tr>
</thead>
<tbody>
<tr>
<td>Modem Reset</td>
<td>9869</td>
<td></td>
<td>Resets the internal modem; used when the modem fails to answer an incoming call or to synchronize with an external modem.</td>
</tr>
<tr>
<td></td>
<td></td>
<td></td>
<td><strong>NOTICE</strong></td>
</tr>
<tr>
<td></td>
<td></td>
<td></td>
<td>Database corruption could occur if the modem is reset during a DB Programming session. Inter-Tel recommends that you assign the modem extension to a Speed Dial button.</td>
</tr>
<tr>
<td>Night Ring On/Off</td>
<td>9860</td>
<td></td>
<td>Places system in night mode or cancels night mode. While in night mode, the night toll and trunk restriction lists are used.</td>
</tr>
<tr>
<td>Periodic Diagnostics On/Off</td>
<td>9825</td>
<td></td>
<td>Periodic diagnostics extends and improves the functionality and checks and reconciliations performed on various resources in the system.</td>
</tr>
<tr>
<td>Program System Speed Dial</td>
<td>9801</td>
<td></td>
<td>Programs system-wide Speed Dial numbers when followed by location codes (000–999).</td>
</tr>
<tr>
<td>Set Network Date/Time</td>
<td>9810</td>
<td></td>
<td>Programs the time and date for all nodes in the system. (Local time and date can also be reset by individual nodes.)</td>
</tr>
<tr>
<td>Set Time/Date</td>
<td>9800</td>
<td></td>
<td>Programs system time and date.</td>
</tr>
<tr>
<td>Synchronize Network Time</td>
<td>9811</td>
<td></td>
<td>Synchronizes the minutes past the hour across the network. If a node’s time is off by more than 30 minutes, synchronizing the minutes may cause the hour to change. Also, network time is automatically synchronized every day at 12:30 AM (00:30), using the time setting on the node with the lowest number.</td>
</tr>
</tbody>
</table>
Diagnostics Feature Codes

The Diagnostics Mode feature code (9900) [9100 in Europe] must be entered at the administrator’s endpoint to enable system diagnostic mode before the feature codes summarized in Table 53 can be used.

### Table 53. Diagnostics Feature Codes

<table>
<thead>
<tr>
<th>Feature Name</th>
<th>Code</th>
<th>Definition</th>
</tr>
</thead>
<tbody>
<tr>
<td>Compression On/Off</td>
<td>9982</td>
<td>Compresses call processing messages sent to DB Programming, speeding up transfers. You should not disable this feature unless instructed to do so by Inter-Tel personnel.</td>
</tr>
<tr>
<td></td>
<td>[9182]</td>
<td></td>
</tr>
<tr>
<td>Compression Statistics</td>
<td>9981</td>
<td>Dumps various statistics related to the DB Programming compression algorithm. This feature should be used only when directed to do so by Inter-Tel personnel.</td>
</tr>
<tr>
<td></td>
<td>[9181]</td>
<td></td>
</tr>
<tr>
<td>Diagnostic – ASAI Snoop Off</td>
<td>9926</td>
<td>Turns off the ASAI output to Message Print. ASAI is the protocol the system uses to talk to DB Programming and the AVDAP. Turning this feature on helps the Inter-Tel Engineers debug the messaging between Call Processing and DB Programming or Call Processing and the AVDAP.</td>
</tr>
<tr>
<td></td>
<td>[9126]</td>
<td></td>
</tr>
<tr>
<td>Diagnostic – ASAI Snoop On</td>
<td>9927</td>
<td>Turns on the ASAI output to Message Print. ASAI is the protocol the system uses to talk to DB Programming and the voice processing system. Turning this feature on helps the Inter-Tel Engineers debug the messaging between Call Processing and DB Programming or Call Processing and the AVDAP.</td>
</tr>
<tr>
<td></td>
<td>[9127]</td>
<td></td>
</tr>
<tr>
<td>Diagnostic – Dump Extension</td>
<td>9922</td>
<td>Allows an Inter-Tel field technician to dump a device or structure when debugging a problem. The system dumps the internal data structures for that extension to message print. If you press the pound button (#) you are prompted for a command. The command allows you to enter an alphanumeric string of the structure to dump.</td>
</tr>
<tr>
<td></td>
<td>[9122]</td>
<td></td>
</tr>
</tbody>
</table>
Table 53. Diagnostics Feature Codes (Continued)

<table>
<thead>
<tr>
<th>Feature Name</th>
<th>Code U.S. [Europe]</th>
<th>Definition</th>
</tr>
</thead>
<tbody>
<tr>
<td>Diagnostic – Dump Node Information</td>
<td>9936 [9136]</td>
<td>The user can easily dump node information to Message Print using a single feature code. Pressing the Dump Node Information feature code dumps specified node information to Message Print for diagnostic purpose. When the system prompts for a node number, enter the applicable node number or zero (0) for all nodes within a network. If the node does not exist, an error message saying INVALID NODE NUMBER appears and prompts the user for a node number again. After entering a node number, the system displays a confirmation message on the endpoint display. This helps for the user to analyze which nodes are up and which nodes are down (the word ‘down’ does not mean that the node is completely down, it simply means the node is unreachable). For Example: If all nodes are up, the display shows ALL NODES ARE UP X. The ‘X’ represents the total number of the nodes. If node 3 and 5 of a 5-node network are down, the display shows # NODES DOWN 2 3 5.</td>
</tr>
<tr>
<td>Diagnostic – Dump System Diagnostics</td>
<td>9933 [9133]</td>
<td>Allow a system administrator to dump system diagnostic information to Message Print.</td>
</tr>
<tr>
<td>Diagnostic – Heap Dump</td>
<td>9943 [9143]</td>
<td>This is used by Inter-Tel software developers and cannot be used in Beta or Production software.</td>
</tr>
<tr>
<td>Diagnostic – Heap Statistics</td>
<td>9947 [9147]</td>
<td>This feature code outputs miscellaneous heap information to Message Print as well as putting up a message on the endpoint that indicates the percentage of available dynamic heap memory. This feature code is useful in determining if the system is losing heap memory and how quickly the system may be losing it.</td>
</tr>
</tbody>
</table>
### Table 53. Diagnostics Feature Codes (Continued)

<table>
<thead>
<tr>
<th>Feature Name</th>
<th>Code U.S. [Europe]</th>
<th>Definition</th>
</tr>
</thead>
<tbody>
<tr>
<td>Diagnostic – ISDN View</td>
<td>9948 [9148]</td>
<td>This feature code is toggles through the three different ISDN view output modes. Entering the feature code the first time puts the ISDN view feature into headers only mode. In this mode the system outputs all ISDN messages to message print in header format (i.e., it does not contain any ISDN information elements). Entering the feature code the second time puts the ISDN view feature into full mode. In this mode the system outputs all ISDN message to message print in full format (i.e., each ISDN information element). Entering the feature code one more time turns this feature off.</td>
</tr>
<tr>
<td>Diagnostic – Major Reset</td>
<td>9962 [9162]</td>
<td>This is used by Inter-Tel software developers and cannot be used in Beta or Production software.</td>
</tr>
<tr>
<td>Diagnostic – Mark As Leaks</td>
<td>9945 [9145]</td>
<td>This is used by Inter-Tel software developers and cannot be used in Beta or Production software.</td>
</tr>
<tr>
<td>Diagnostic - Mark As Quiescent</td>
<td>9946 [9146]</td>
<td>This is used by Inter-Tel software developers and cannot be used in Beta or Production software.</td>
</tr>
<tr>
<td>Diagnostic – Minor Reset</td>
<td>9964 [9164]</td>
<td>This is used by Inter-Tel software developers and cannot be used in Beta or Production software.</td>
</tr>
<tr>
<td>Diagnostic – Network Freeze Zone System Histories</td>
<td>9939 [9139]</td>
<td>The system fault history for any freeze zone in the network can be halted (frozen) or re-enabled using these feature codes when diagnostics mode is enabled. The fault history can then be extracted from each zone and used by service personnel when troubleshooting the system.</td>
</tr>
<tr>
<td>Diagnostic – Network Unfreeze Zone System Histories</td>
<td>9989 [9189]</td>
<td></td>
</tr>
<tr>
<td>Diagnostic – Network Groups</td>
<td>9963 [9163]</td>
<td>Allows an administrator to verify that the Network Groups on the local node are programmed properly. When 9963 is entered at an administrator’s endpoint, the telephone system initiates pings from each Inter-Tel IP device on the local node and determines if the other devices respond to the ping. If a device does not respond to the ping or if a firewall is detected, the system issues a Message Print message.</td>
</tr>
<tr>
<td>Diagnostic – Print Auxdata</td>
<td>9972 [9172]</td>
<td>(Not Programmable) Sends a report to a designated printer or file that shows system reset history information to be used for troubleshooting purposes.</td>
</tr>
<tr>
<td>Feature Name</td>
<td>Code U.S. [Europe]</td>
<td>Definition</td>
</tr>
<tr>
<td>----------------------------</td>
<td>-------------------</td>
<td>------------------------------------------------------------------------------------------------------------------------------------------</td>
</tr>
<tr>
<td>Diagnostic – Print Message Log</td>
<td>9975 [9175]</td>
<td>(Not Programmable) Sends a report to a designated printer or file that lists system error messages to be used for troubleshooting purposes.</td>
</tr>
<tr>
<td>Diagnostic – Print Network Log</td>
<td>9976 [9176]</td>
<td>This feature code prints the network log to message print. This feature is useful in determining the system of a networked system.</td>
</tr>
<tr>
<td>Diagnostic – Query Node Traffic</td>
<td>9978 [9178]</td>
<td>Using this feature code you can query the status of various devices on the system based on the status of the traffic flags in DB Programming. The output is sent to Message Print.</td>
</tr>
<tr>
<td>Diagnostic – Show Version</td>
<td>9928 [9128]</td>
<td>To check the call processing software version at an administrator’s endpoint, you can enable diagnostics mode, then enter this feature code to view the version and date of the call processing software. However, feature code 9928 displays the firmware version of the endpoint if it is in SIP mode.</td>
</tr>
<tr>
<td>Diagnostic – SIP View</td>
<td>9987 [9187]</td>
<td>Allows the user to change the system wide SIP output value. Options include No Output, Headers and Full Output.</td>
</tr>
<tr>
<td>Diagnostic – Spare 1–3</td>
<td>9910–9912 [9110–9112]</td>
<td>This is used by Inter-Tel software developers and cannot be used in Beta or Production software.</td>
</tr>
<tr>
<td>Diagnostic – System History</td>
<td>9974 [9174]</td>
<td>This is used by Inter-Tel software developers and cannot be used in Beta or Production software.</td>
</tr>
<tr>
<td>Diagnostic – View Displays</td>
<td>9983 [9183]</td>
<td>This is used by Inter-Tel software developers and cannot be used in Beta or Production software.</td>
</tr>
<tr>
<td>Program Database</td>
<td>9932 [9132]</td>
<td>Can be used for programming endpoint, system, and trunk parameters.</td>
</tr>
<tr>
<td>Seize Device</td>
<td>9973 [9173]</td>
<td>Used during troubleshooting to seize a specific trunk or endpoint by indicating the board number, port number, and device number.</td>
</tr>
<tr>
<td>System History – Freeze</td>
<td>9993 [9193]</td>
<td>The system fault history can be halted (frozen) or re-enabled using these feature codes when diagnostics mode is enabled. Fault history is used by service personnel when troubleshooting the system.</td>
</tr>
<tr>
<td>System History – Unfreeze</td>
<td>9998 [9198]</td>
<td></td>
</tr>
</tbody>
</table>
Default Voice Processing Prompts

This section contains the default voice prompts for the voice processing systems supported by the Inter-Tel 5000 platform. The Inter-Tel platform supports the following voice processing systems:

- **Basic Voice Mail**: The built-in voice processing system, Basic Voice Mail (BVM) provides voice mail and voice messaging for system users. BVM runs on the Linux® operating system and does not support Fax-On-Demand, buffered SMDR, or Unified Messaging. If part of a network, mailboxes on an Inter-Tel 5000 node equipped only with BVM cannot forward messages to, or receive messages forwarded from, mailboxes on other nodes.

- **Enterprise® Messaging**: Enterprise Messaging (EM) runs on the Windows 2000 operating system. EM is Inter-Tel’s the newest voice processing system. EM unifies communication and messaging by providing access to voice mail, fax, and e-mail messages in the user’s mailbox. In addition, EM provides automatic speech recognition capabilities and built-in Unified Messaging features. For detailed information about EM, see the *Enterprise Messaging Installation Manual* (part no. 780.8006).

- **Voice Processing Unit**: The Voice Processing Unit (VPU) runs on either the Windows NT or Windows 2000 operating systems. The VPU system is installed on an IBM-compatible PC that is connected to the Inter-Tel system to provide integrated voice processing features such as Voice Mail, Directories, Record-A-Call, etc.

**Automatic Speech Recognition (ASR)**

The Enterprise Messaging (EM) voice processing system supports Automatic Speech Recognition (ASR), which allows users to access voice mail, call routing applications, the automated attendant and the directories by issuing spoken commands instead of dialing digits on their endpoint. If ASR is enabled on a voice processing system, users are prompted to speak or press a digit on the dialpad to complete a voice mail task. If ASR is not enabled, users are only prompted to press a digit. ASR is currently supported on EM voice processing systems only.
Default ASR and Non-ASR Voice Prompts

Table 54 shows all of the U.S. English default system voice prompts for ASR- and non ASR-enabled systems. Prompts designated by an asterisk (*) cannot be re-recorded.

Table 54: Voice Prompts: Non-ASR and ASR

<table>
<thead>
<tr>
<th>Prompt ID</th>
<th>Content (Non-ASR)</th>
<th>Content (ASR)</th>
</tr>
</thead>
<tbody>
<tr>
<td>001</td>
<td>To leave a message, please enter a mailbox number.</td>
<td>To leave a message, please say or enter a mailbox number.</td>
</tr>
<tr>
<td>002</td>
<td>For the directory, press pound (or hash in European systems).</td>
<td>For the directory, say “Directory” or press pound (or hash).</td>
</tr>
<tr>
<td>003</td>
<td>To reach an operator, press zero.</td>
<td>To reach an operator, say “Operator” or press zero.</td>
</tr>
<tr>
<td>004</td>
<td>I’m sorry...</td>
<td>N/A</td>
</tr>
<tr>
<td>005</td>
<td>...is not available.</td>
<td>N/A</td>
</tr>
<tr>
<td>006</td>
<td>The number entered is not valid.</td>
<td>N/A</td>
</tr>
<tr>
<td>007</td>
<td>Using the keys on your telephone, please spell the name of the person, then press pound (or hash). For the letter Q, press seven. For the letter Z, press nine. To advance to the next character, please wait for two seconds or press another key. To exit the directory, press star.</td>
<td>Please say the name of the person you are trying to reach. Otherwise, using the keys on your telephone, please spell the last name of the person, then press pound (or hash). For the letter Q, press seven. For the letter Z, press nine. To advance to the next character, please wait for two seconds or press another key. To exit the directory, press star.</td>
</tr>
<tr>
<td>008</td>
<td>To exit the directory, press star.</td>
<td>To exit the directory, say “Cancel” or press star.</td>
</tr>
<tr>
<td>009</td>
<td>Mailbox number...</td>
<td>N/A</td>
</tr>
<tr>
<td>010</td>
<td>If this is correct, press pound (or hash).</td>
<td>If this is correct, say “Yes” or press pound (or hash).</td>
</tr>
<tr>
<td>011</td>
<td>For the previous name in the directory, press one.</td>
<td>For the previous name, say “Previous” or press one.</td>
</tr>
<tr>
<td>012</td>
<td>To hear additional information, press two.</td>
<td>To hear additional information, say “More” or press two.</td>
</tr>
<tr>
<td>013</td>
<td>To hear the next directory name, press three.</td>
<td>To hear the next name, say “Next” or press three</td>
</tr>
<tr>
<td>014</td>
<td>To spell a new name, press four.</td>
<td>To spell a new name, say “New” or press four.</td>
</tr>
<tr>
<td>015</td>
<td>Extension number...</td>
<td>N/A</td>
</tr>
<tr>
<td>016</td>
<td>Not used</td>
<td>Not used</td>
</tr>
<tr>
<td>017</td>
<td>...is not a valid choice.</td>
<td>N/A</td>
</tr>
<tr>
<td>018</td>
<td>Please try again.</td>
<td>N/A</td>
</tr>
<tr>
<td>019</td>
<td>After the tone, please record your message. When finished, you may hang up to deliver the message or press pound (or hash) for more options.</td>
<td>N/A</td>
</tr>
<tr>
<td>020</td>
<td>Message delivered.</td>
<td>N/A</td>
</tr>
<tr>
<td>021</td>
<td>To replay, press one.</td>
<td>To replay, say “Replay” or press one.</td>
</tr>
<tr>
<td>022</td>
<td>To erase and re-record, press three.</td>
<td>To erase and re-record, say “Erase” or press three.</td>
</tr>
</tbody>
</table>
### Table 54: Voice Prompts: Non-ASR and ASR (Continued)

<table>
<thead>
<tr>
<th>Prompt ID</th>
<th>Content (Non-ASR)</th>
<th>Content (ASR)</th>
</tr>
</thead>
<tbody>
<tr>
<td>023</td>
<td>To append this recording, press two.</td>
<td>To append this recording, say “Append” or press two.</td>
</tr>
<tr>
<td>024</td>
<td>To accept, press pound (or hash).</td>
<td>To accept, say “Accept” or press pound (or hash).</td>
</tr>
<tr>
<td>025</td>
<td>Please record after the tone. To end, press pound (or hash).</td>
<td>N/A</td>
</tr>
<tr>
<td>026</td>
<td>Recording erased.</td>
<td>N/A</td>
</tr>
<tr>
<td>027</td>
<td>Recording stopped.</td>
<td>N/A</td>
</tr>
<tr>
<td>028</td>
<td>Please begin speaking after the tone.</td>
<td>N/A</td>
</tr>
<tr>
<td>029</td>
<td>Message play stopped. To continue, press pound (or hash).</td>
<td>Message play stopped. To continue, say “Resume” or press pound (or hash).</td>
</tr>
<tr>
<td>030</td>
<td>Enter the destination mailbox or group list number.</td>
<td>Say or enter the destination mailbox or group list number.</td>
</tr>
<tr>
<td>031</td>
<td>Logon canceled.</td>
<td>N/A</td>
</tr>
<tr>
<td>032</td>
<td>Enter your password. Then press pound (or hash).</td>
<td>Say or enter your password.</td>
</tr>
<tr>
<td>033</td>
<td>There have been three invalid attempts to log on to your mailbox.</td>
<td>N/A</td>
</tr>
<tr>
<td>034</td>
<td>Invalid password.</td>
<td>N/A</td>
</tr>
<tr>
<td>035</td>
<td>Thank you for calling. Good-bye.</td>
<td>N/A</td>
</tr>
<tr>
<td>036</td>
<td>To access mailbox features, you must change your system-assigned password.</td>
<td>N/A</td>
</tr>
<tr>
<td>037</td>
<td>You have not finished setting up your mailbox.</td>
<td>N/A</td>
</tr>
<tr>
<td>038</td>
<td>Subscriber access denied.</td>
<td>N/A</td>
</tr>
<tr>
<td>039</td>
<td>Because this is the first time you are entering your mailbox, you must record your name for the company directory.</td>
<td>N/A</td>
</tr>
<tr>
<td>040</td>
<td>Since you are entering your mailbox for the first time, you should be aware of a few standard features. To cancel an option you have accessed and return to the previous menu, press star. To accept your recording or dialing sequence, or to advance to your next message, press pound (or hash). For help, you may press zero at any time you are at a menu and the system will replay the current menu and/or give you additional information. Certain mailbox features such as remote messaging are available only if authorized by your System Administrator. Your mailbox is now fully set up.</td>
<td>Since you are entering your mailbox for the first time, you should be aware of a few standard features. To cancel an option you have accessed and return to the previous menu, say “Cancel” or press star. To accept your recording or dialing sequence, or to advance to your next message, say “Accept” or press pound (or hash). For help, say “Help” or press zero to replay the current menu and/or give you additional information. Certain mailbox features such as remote messaging are available only if authorized by your System Administrator. Your mailbox is now fully set up.</td>
</tr>
<tr>
<td>041</td>
<td>Your password may be up to 12 digits.</td>
<td>N/A</td>
</tr>
<tr>
<td>042</td>
<td>Enter your new password, then press pound (or hash). If you don’t want a password, press pound (or hash) now.</td>
<td>Say or enter your new password. If you don’t want a password, say “Skip” or press pound (or hash).</td>
</tr>
<tr>
<td>043</td>
<td>Your new password is…</td>
<td>N/A</td>
</tr>
</tbody>
</table>
Table 54: Voice Prompts: Non-ASR and ASR (Continued)

<table>
<thead>
<tr>
<th>Prompt ID</th>
<th>Content (Non-ASR)</th>
<th>Content (ASR)</th>
</tr>
</thead>
<tbody>
<tr>
<td>044</td>
<td>The number entered is too long.</td>
<td>N/A</td>
</tr>
<tr>
<td>045</td>
<td>To accept this entry, press pound (or hash).</td>
<td>To accept this entry, say “Accept” or press pound (or hash).</td>
</tr>
<tr>
<td>046</td>
<td>To erase and re-enter, press three.</td>
<td>To erase and re-enter, say “Re-enter” or press three.</td>
</tr>
<tr>
<td>047</td>
<td>Password erased.</td>
<td>N/A</td>
</tr>
<tr>
<td>048</td>
<td>Password saved.</td>
<td>N/A</td>
</tr>
<tr>
<td>049</td>
<td>Your mailbox is full and cannot receive any new messages.</td>
<td>N/A</td>
</tr>
<tr>
<td>050 - 051</td>
<td>Not used</td>
<td>Not used</td>
</tr>
<tr>
<td>052</td>
<td>Your mailbox is almost full. Please delete any messages you no longer need so new messages may arrive.</td>
<td>N/A</td>
</tr>
<tr>
<td>053</td>
<td>You have...</td>
<td>N/A</td>
</tr>
<tr>
<td>054</td>
<td>You have one new message.</td>
<td>N/A</td>
</tr>
<tr>
<td>055</td>
<td>...new messages.</td>
<td>N/A</td>
</tr>
<tr>
<td>056</td>
<td>...of which is priority.</td>
<td>N/A</td>
</tr>
<tr>
<td>057</td>
<td>You have no new messages.</td>
<td>N/A</td>
</tr>
<tr>
<td>058</td>
<td>You have one saved message.</td>
<td>N/A</td>
</tr>
<tr>
<td>059</td>
<td>...saved messages.</td>
<td>N/A</td>
</tr>
<tr>
<td>060</td>
<td>Remote messaging is currently enabled.</td>
<td>N/A</td>
</tr>
<tr>
<td>061</td>
<td>To listen to new messages, press one.</td>
<td>To listen to new messages, say “New” or press one.</td>
</tr>
<tr>
<td>062</td>
<td>To record a message, press two.</td>
<td>To record a message, say “Record” or press two.</td>
</tr>
<tr>
<td>063</td>
<td>To listen to saved messages, press three.</td>
<td>To listen to saved messages, say “Saved” or press three.</td>
</tr>
<tr>
<td>064</td>
<td>For personal options, press four.</td>
<td>For personal options, say “Personal” or press four.</td>
</tr>
<tr>
<td>065</td>
<td>To setup remote messaging, press five.</td>
<td>To setup remote messaging, say “Remote” or press five.</td>
</tr>
<tr>
<td>066</td>
<td>For your primary setup, press one.</td>
<td>For your primary setup, say “Primary” or press one.</td>
</tr>
<tr>
<td>067</td>
<td>For your alternate setup, press two.</td>
<td>For your alternate setup, say “Alternate” or press two.</td>
</tr>
<tr>
<td>068</td>
<td>If this is the correct destination, press pound (or hash).</td>
<td>If this is the correct destination, say “Yes” or press pound (or hash).</td>
</tr>
<tr>
<td>069</td>
<td>To send this message now, press pound (or hash).</td>
<td>To send this message now, say “Send” or press pound (or hash).</td>
</tr>
<tr>
<td>070</td>
<td>For special delivery options, press nine.</td>
<td>For special delivery options, say “Options” or press nine.</td>
</tr>
<tr>
<td>071</td>
<td>To return to the previous menu, press star.</td>
<td>To return to the previous menu, say “Cancel” or press star.</td>
</tr>
</tbody>
</table>
### Table 54: Voice Prompts: Non-ASR and ASR (Continued)

<table>
<thead>
<tr>
<th>Prompt ID</th>
<th>Content (Non-ASR)</th>
<th>Content (ASR)</th>
</tr>
</thead>
<tbody>
<tr>
<td>072</td>
<td>Message canceled.</td>
<td>N/A</td>
</tr>
<tr>
<td>073</td>
<td>To mark this message private, press one.</td>
<td>To mark this message private, say “Private” or press one.</td>
</tr>
<tr>
<td>074</td>
<td>To mark this message for certified delivery, press two.</td>
<td>To mark this message for certified delivery, say “Certified” or press two.</td>
</tr>
<tr>
<td>075</td>
<td>To mark this message priority, press three.</td>
<td>To mark this message priority, say “Priority” or press three.</td>
</tr>
<tr>
<td>076</td>
<td>Message marked private.</td>
<td>N/A</td>
</tr>
<tr>
<td>077</td>
<td>Message marked certified.</td>
<td>N/A</td>
</tr>
<tr>
<td>078</td>
<td>Message marked priority.</td>
<td>N/A</td>
</tr>
<tr>
<td>079</td>
<td>To send this message to additional mailboxes, press pound (or hash).</td>
<td>To send this message to additional mailboxes, say “Yes” or press pound (or hash).</td>
</tr>
<tr>
<td>080</td>
<td>To rewind your message or recording, press one.</td>
<td>To rewind your message or recording, say “Backup” or press one.</td>
</tr>
<tr>
<td>081</td>
<td>To pause at any time, press two.</td>
<td>To pause at any time, say “Pause” or press two.</td>
</tr>
<tr>
<td>082</td>
<td>To fast forward your message or recording, press three.</td>
<td>To fast forward your message or recording say “Forward” or press three.</td>
</tr>
<tr>
<td>083</td>
<td>To lower the volume, press four.</td>
<td>To lower the volume, say “Lower” or press four.</td>
</tr>
<tr>
<td>084</td>
<td>To raise the volume, press six.</td>
<td>To raise the volume, say “Higher” or press six.</td>
</tr>
<tr>
<td>085</td>
<td>To save this message, press seven.</td>
<td>To save this message, say “Save” or press seven.</td>
</tr>
<tr>
<td>086</td>
<td>To delete this message, press nine.</td>
<td>To delete this message, say “Delete Message” or press nine.</td>
</tr>
<tr>
<td>087</td>
<td>To skip to the end of your message or recording, press pound (or hash).</td>
<td>To skip to the end of your message or recording, say “Skip” or press pound (or hash).</td>
</tr>
<tr>
<td>088</td>
<td>To return to the main menu, press star.</td>
<td>To return to the main menu, say “Cancel” or press star.</td>
</tr>
<tr>
<td>089</td>
<td>To reply to this message, press two.</td>
<td>To reply to this message, say “Reply” or press two.</td>
</tr>
<tr>
<td>090</td>
<td>To forward a copy of this message to another destination, press three.</td>
<td>To forward a copy of this message to another destination, say “Forward” or press three.</td>
</tr>
<tr>
<td>091</td>
<td>To replay the previous message, press four.</td>
<td>To replay the previous message, say “Previous” or press four.</td>
</tr>
<tr>
<td>092</td>
<td>To replay the message envelope, press five.</td>
<td>To replay the message envelope, say “Envelope” or press five.</td>
</tr>
<tr>
<td>093</td>
<td>To listen to your next message, press six.</td>
<td>To listen to your next message, say “Next” or press six.</td>
</tr>
<tr>
<td>094</td>
<td>There are no previous messages.</td>
<td>N/A</td>
</tr>
<tr>
<td>095</td>
<td>Message saved.</td>
<td>N/A</td>
</tr>
<tr>
<td>096</td>
<td>Message deleted.</td>
<td>N/A</td>
</tr>
<tr>
<td>097</td>
<td>There are no further messages.</td>
<td>N/A</td>
</tr>
<tr>
<td>098</td>
<td>Message received...</td>
<td>N/A</td>
</tr>
</tbody>
</table>
## Table 54: Voice Prompts: Non-ASR and ASR (Continued)

<table>
<thead>
<tr>
<th>Prompt ID</th>
<th>Content (Non-ASR)</th>
<th>Content (ASR)</th>
</tr>
</thead>
<tbody>
<tr>
<td>099</td>
<td>...at...</td>
<td>N/A</td>
</tr>
<tr>
<td>100</td>
<td>...from...</td>
<td>N/A</td>
</tr>
<tr>
<td>101</td>
<td>...totaling...</td>
<td>N/A</td>
</tr>
<tr>
<td>102</td>
<td>...minutes...</td>
<td>N/A</td>
</tr>
<tr>
<td>103</td>
<td>...minute...</td>
<td>N/A</td>
</tr>
<tr>
<td>104*</td>
<td>...seconds...</td>
<td>N/A</td>
</tr>
<tr>
<td>105</td>
<td>...second...</td>
<td>N/A</td>
</tr>
<tr>
<td>106</td>
<td>Message received from an outside caller.</td>
<td>N/A</td>
</tr>
<tr>
<td>107</td>
<td>To record an introduction with a copy of this message, press one.</td>
<td>To record an introduction with a copy of this message, say “Record” or press one.</td>
</tr>
<tr>
<td>108</td>
<td>To change your personal greeting, press one.</td>
<td>To change your personal greeting, say “Greeting” or press one.</td>
</tr>
<tr>
<td>109</td>
<td>To change your name in the directory, press two.</td>
<td>To change your name in the directory, say “Name” or press two.</td>
</tr>
<tr>
<td>110</td>
<td>To change your password, press three.</td>
<td>To change your password, say “Password” or press three.</td>
</tr>
<tr>
<td>111</td>
<td>To change your message envelope, press four.</td>
<td>To change your message envelope, say “Envelope” or press four.</td>
</tr>
<tr>
<td>112</td>
<td>To replay the message, press one.</td>
<td>To replay the message, say “Replay” or press one.</td>
</tr>
<tr>
<td>113</td>
<td>To select your primary greeting, press one.</td>
<td>To select your primary greeting, say “Primary” or press one.</td>
</tr>
<tr>
<td>114</td>
<td>To select your alternate greeting, press two.</td>
<td>To select your alternate greeting, say “Alternate” or press two.</td>
</tr>
<tr>
<td>115</td>
<td>To select the system greeting, press three.</td>
<td>To select the system greeting, say “System” or press three.</td>
</tr>
<tr>
<td>116</td>
<td>After the tone, please record your greeting, then press pound (or hash).</td>
<td>N/A</td>
</tr>
<tr>
<td>117</td>
<td>Greeting saved.</td>
<td>N/A</td>
</tr>
<tr>
<td>118</td>
<td>After the tone, please record your name, then press pound (or hash).</td>
<td>N/A</td>
</tr>
<tr>
<td>119</td>
<td>Name saved.</td>
<td>N/A</td>
</tr>
<tr>
<td>120</td>
<td>Your message envelope is currently set to return...</td>
<td>N/A</td>
</tr>
<tr>
<td>121</td>
<td>...all information.</td>
<td>N/A</td>
</tr>
<tr>
<td>122</td>
<td>...time and date...</td>
<td>N/A</td>
</tr>
<tr>
<td>123</td>
<td>...message source...</td>
<td>N/A</td>
</tr>
<tr>
<td>124*</td>
<td>...and...</td>
<td>N/A</td>
</tr>
<tr>
<td>125</td>
<td>...message length...</td>
<td>N/A</td>
</tr>
</tbody>
</table>
### Table 54: Voice Prompts: Non-ASR and ASR (Continued)

<table>
<thead>
<tr>
<th>Prompt ID</th>
<th>Content (Non-ASR)</th>
<th>Content (ASR)</th>
</tr>
</thead>
<tbody>
<tr>
<td>126</td>
<td>To turn on all envelope options, press four.</td>
<td>To turn on all envelope options, say “All Options” or press four.</td>
</tr>
<tr>
<td>127</td>
<td>For message source, press two.</td>
<td>For message source, say “Source” or press two.</td>
</tr>
<tr>
<td>128</td>
<td>For message length, press three.</td>
<td>For message length, say “Length” or press three.</td>
</tr>
<tr>
<td>129</td>
<td>This option is now off.</td>
<td>N/A</td>
</tr>
<tr>
<td>130</td>
<td>This option is now on.</td>
<td>N/A</td>
</tr>
<tr>
<td>131</td>
<td>For time and date, press one.</td>
<td>For time and date, say “Time, Date” or press one.</td>
</tr>
<tr>
<td>132</td>
<td>Your message envelope will play all options.</td>
<td>N/A</td>
</tr>
<tr>
<td>133</td>
<td>...is turned on to call you at...</td>
<td>N/A</td>
</tr>
<tr>
<td>134*</td>
<td>AM</td>
<td>N/A</td>
</tr>
<tr>
<td>135*</td>
<td>PM</td>
<td>N/A</td>
</tr>
<tr>
<td>136</td>
<td>...Extension ID number...</td>
<td>N/A</td>
</tr>
<tr>
<td>137</td>
<td>...Every day.</td>
<td>N/A</td>
</tr>
<tr>
<td>138</td>
<td>...Monday through Friday.</td>
<td>N/A</td>
</tr>
<tr>
<td>139*</td>
<td>Sunday</td>
<td>N/A</td>
</tr>
<tr>
<td>140*</td>
<td>Monday</td>
<td>N/A</td>
</tr>
<tr>
<td>141*</td>
<td>Tuesday</td>
<td>N/A</td>
</tr>
<tr>
<td>142*</td>
<td>Wednesday</td>
<td>N/A</td>
</tr>
<tr>
<td>143*</td>
<td>Thursday</td>
<td>N/A</td>
</tr>
<tr>
<td>144*</td>
<td>Friday</td>
<td>N/A</td>
</tr>
<tr>
<td>145*</td>
<td>Saturday</td>
<td>N/A</td>
</tr>
<tr>
<td>146</td>
<td>To turn this number on or off, press one.</td>
<td>To turn this number on or off, say “Change” or press one.</td>
</tr>
<tr>
<td>147</td>
<td>To change the phone number, press two.</td>
<td>To change the phone number, say “Number” or press two.</td>
</tr>
<tr>
<td>148</td>
<td>To change times of day, press two.</td>
<td>To change times of day, say “Time” or press two.</td>
</tr>
<tr>
<td>149</td>
<td>To change days of the week, press three.</td>
<td>To change days of the week, say “Days” or press three.</td>
</tr>
<tr>
<td>150</td>
<td>For pager notification, press one.</td>
<td>For pager notification, say “Pager” or press one.</td>
</tr>
<tr>
<td>151</td>
<td>For personal phone notification, press two.</td>
<td>For personal phone notification, say “Personal” or press two.</td>
</tr>
<tr>
<td>152</td>
<td>For notification of all new messages, press one.</td>
<td>For notification of all new messages, say “All Messages” or press one.</td>
</tr>
<tr>
<td>153</td>
<td>For priority-only, press two.</td>
<td>For priority-only, say “Priority” or press two.</td>
</tr>
<tr>
<td>154</td>
<td>Please enter the destination telephone number and wait for further instructions.</td>
<td>Please say or enter the destination telephone number and wait for further instructions.</td>
</tr>
<tr>
<td>155</td>
<td>The new number is...</td>
<td>N/A</td>
</tr>
<tr>
<td>156</td>
<td>Please enter the start time.</td>
<td>Please say or enter the start time.</td>
</tr>
</tbody>
</table>
### Table 54: Voice Prompts: Non-ASR and ASR (Continued)

<table>
<thead>
<tr>
<th>Prompt ID</th>
<th>Content (Non-ASR)</th>
<th>Content (ASR)</th>
</tr>
</thead>
<tbody>
<tr>
<td>157</td>
<td>Please enter the stop time.</td>
<td>Please say or enter the stop time.</td>
</tr>
<tr>
<td>158</td>
<td>I’m sorry. You must program a telephone number first.</td>
<td>N/A</td>
</tr>
<tr>
<td>159</td>
<td>Invalid time.</td>
<td>N/A</td>
</tr>
<tr>
<td>160</td>
<td>For AM, press one.</td>
<td>For AM, say “AM” or press one.</td>
</tr>
<tr>
<td>161</td>
<td>Please select the days of the week on which you wish to be notified.</td>
<td>N/A</td>
</tr>
<tr>
<td>162</td>
<td>Not used</td>
<td>Not used</td>
</tr>
<tr>
<td>163</td>
<td>Please enter your mailbox number.</td>
<td>Please say or enter your mailbox number.</td>
</tr>
<tr>
<td>164</td>
<td>Today</td>
<td>N/A</td>
</tr>
<tr>
<td>165</td>
<td>Yesterday</td>
<td>N/A</td>
</tr>
<tr>
<td>166*</td>
<td>January</td>
<td>N/A</td>
</tr>
<tr>
<td>167*</td>
<td>February</td>
<td>N/A</td>
</tr>
<tr>
<td>168*</td>
<td>March</td>
<td>N/A</td>
</tr>
<tr>
<td>169*</td>
<td>April</td>
<td>N/A</td>
</tr>
<tr>
<td>170*</td>
<td>May</td>
<td>N/A</td>
</tr>
<tr>
<td>171*</td>
<td>June</td>
<td>N/A</td>
</tr>
<tr>
<td>172*</td>
<td>July</td>
<td>N/A</td>
</tr>
<tr>
<td>173*</td>
<td>August</td>
<td>N/A</td>
</tr>
<tr>
<td>174*</td>
<td>September</td>
<td>N/A</td>
</tr>
<tr>
<td>175*</td>
<td>October</td>
<td>N/A</td>
</tr>
<tr>
<td>176*</td>
<td>November</td>
<td>N/A</td>
</tr>
<tr>
<td>177*</td>
<td>December</td>
<td>N/A</td>
</tr>
<tr>
<td>178*</td>
<td>First</td>
<td>N/A</td>
</tr>
<tr>
<td>179*</td>
<td>Second</td>
<td>N/A</td>
</tr>
<tr>
<td>180*</td>
<td>Third</td>
<td>N/A</td>
</tr>
<tr>
<td>181*</td>
<td>Fourth</td>
<td>N/A</td>
</tr>
<tr>
<td>182*</td>
<td>Fifth</td>
<td>N/A</td>
</tr>
<tr>
<td>183*</td>
<td>Sixth</td>
<td>N/A</td>
</tr>
<tr>
<td>184*</td>
<td>Seventh</td>
<td>N/A</td>
</tr>
<tr>
<td>185*</td>
<td>Eighth</td>
<td>N/A</td>
</tr>
<tr>
<td>186*</td>
<td>Ninth</td>
<td>N/A</td>
</tr>
<tr>
<td>187*</td>
<td>Tenth</td>
<td>N/A</td>
</tr>
<tr>
<td>188*</td>
<td>Eleventh</td>
<td>N/A</td>
</tr>
<tr>
<td>189*</td>
<td>Twelfth</td>
<td>N/A</td>
</tr>
<tr>
<td>190*</td>
<td>Thirteenth</td>
<td>N/A</td>
</tr>
</tbody>
</table>
Table 54: Voice Prompts: Non-ASR and ASR (Continued)

<table>
<thead>
<tr>
<th>Prompt ID</th>
<th>Content (Non-ASR)</th>
<th>Content (ASR)</th>
</tr>
</thead>
<tbody>
<tr>
<td>191*</td>
<td>Fourteenth</td>
<td>N/A</td>
</tr>
<tr>
<td>192*</td>
<td>Fifteenth</td>
<td>N/A</td>
</tr>
<tr>
<td>193*</td>
<td>Sixteenth</td>
<td>N/A</td>
</tr>
<tr>
<td>194*</td>
<td>Seventeenth</td>
<td>N/A</td>
</tr>
<tr>
<td>195*</td>
<td>Eighteenth</td>
<td>N/A</td>
</tr>
<tr>
<td>196*</td>
<td>Nineteenth</td>
<td>N/A</td>
</tr>
<tr>
<td>197*</td>
<td>Twentieth</td>
<td>N/A</td>
</tr>
<tr>
<td>198*</td>
<td>Twenty-first</td>
<td>N/A</td>
</tr>
<tr>
<td>199*</td>
<td>Twenty-second</td>
<td>N/A</td>
</tr>
<tr>
<td>200*</td>
<td>Twenty-third</td>
<td>N/A</td>
</tr>
<tr>
<td>201*</td>
<td>Twenty-fourth</td>
<td>N/A</td>
</tr>
<tr>
<td>202*</td>
<td>Twenty-fifth</td>
<td>N/A</td>
</tr>
<tr>
<td>203*</td>
<td>Twenty-sixth</td>
<td>N/A</td>
</tr>
<tr>
<td>204*</td>
<td>Twenty-seventh</td>
<td>N/A</td>
</tr>
<tr>
<td>205*</td>
<td>Twenty-eighth</td>
<td>N/A</td>
</tr>
<tr>
<td>206*</td>
<td>Twenty-ninth</td>
<td>N/A</td>
</tr>
<tr>
<td>207*</td>
<td>Thirtieth</td>
<td>N/A</td>
</tr>
<tr>
<td>208*</td>
<td>Thirty-first</td>
<td>N/A</td>
</tr>
<tr>
<td>209</td>
<td>...the Voice Mail system.</td>
<td>N/A</td>
</tr>
<tr>
<td>210</td>
<td>The system is set to notify you between...</td>
<td>N/A</td>
</tr>
<tr>
<td>211</td>
<td>...is turned off.</td>
<td>N/A</td>
</tr>
<tr>
<td>212</td>
<td>...is turned on to call your pager at...</td>
<td>N/A</td>
</tr>
<tr>
<td>213</td>
<td>...with all new messages.</td>
<td>N/A</td>
</tr>
<tr>
<td>214</td>
<td>...with all priority messages.</td>
<td>N/A</td>
</tr>
<tr>
<td>215</td>
<td>...starting at...</td>
<td>N/A</td>
</tr>
<tr>
<td>216</td>
<td>...ending at...</td>
<td>N/A</td>
</tr>
<tr>
<td>217</td>
<td>For mailbox and group list maintenance, press two.</td>
<td>For mailbox and group list maintenance, say &quot;Mailbox Maintenance&quot; or press two.</td>
</tr>
<tr>
<td>218</td>
<td>To send a message to all subscribers, press one.</td>
<td>To send a message to all subscribers, say &quot;Broadcast Message&quot; or press one.</td>
</tr>
<tr>
<td>219</td>
<td>To make a custom recording, press one.</td>
<td>To make a custom recording, say &quot;Custom&quot; or press one.</td>
</tr>
<tr>
<td>220</td>
<td>To select a custom recording, press two.</td>
<td>To select a custom recording, say &quot;Select&quot; or press two.</td>
</tr>
<tr>
<td>221</td>
<td>Not used</td>
<td>Not used</td>
</tr>
<tr>
<td>222</td>
<td>Please enter the recording number.</td>
<td>N/A</td>
</tr>
</tbody>
</table>
Table 54: Voice Prompts: Non-ASR and ASR (Continued)

<table>
<thead>
<tr>
<th>Prompt ID</th>
<th>Content (Non-ASR)</th>
<th>Content (ASR)</th>
</tr>
</thead>
<tbody>
<tr>
<td>223</td>
<td>After the tone, please record. Then press pound (or hash).</td>
<td>N/A</td>
</tr>
<tr>
<td>224</td>
<td>Recording saved.</td>
<td>N/A</td>
</tr>
<tr>
<td>225</td>
<td>To continue recording, press pound (or hash).</td>
<td>To continue recording, say “Continue” or press pound (or hash).</td>
</tr>
<tr>
<td>226</td>
<td>Not used</td>
<td>Not used</td>
</tr>
<tr>
<td>227</td>
<td>You have no saved messages.</td>
<td>N/A</td>
</tr>
<tr>
<td>228</td>
<td>To spell the first name, press five.</td>
<td>To search by first name, say “First” or press five.</td>
</tr>
<tr>
<td>229</td>
<td>To spell the last name, press five.</td>
<td>To search by last name, say “Last” or press five.</td>
</tr>
<tr>
<td>230</td>
<td>Using the keys on your telephone, please spell the first name of the person. Then press pound (or hash). For the letter Q, press seven. For the letter Z, press nine.</td>
<td>Please say the first name of the person you are trying to reach. Otherwise, using the keys on your telephone, please spell the first name of the person, then press pound (or hash). For the letter Q, press seven. For the letter Z, press nine.</td>
</tr>
<tr>
<td>231</td>
<td>No message left.</td>
<td>N/A</td>
</tr>
<tr>
<td>232</td>
<td>...last...</td>
<td>N/A</td>
</tr>
<tr>
<td>233</td>
<td>...an outside caller at...</td>
<td>N/A</td>
</tr>
<tr>
<td>234</td>
<td>Message was sent from...</td>
<td>N/A</td>
</tr>
<tr>
<td>235</td>
<td>This matches the mailbox of...</td>
<td>N/A</td>
</tr>
<tr>
<td>236</td>
<td>...an unknown caller.</td>
<td>N/A</td>
</tr>
<tr>
<td>237</td>
<td>To turn off the envelope, press five.</td>
<td>To turn off the envelope, say “None” or press five.</td>
</tr>
<tr>
<td>238</td>
<td>To accept the changes, press pound (or hash).</td>
<td>To accept the changes, say “Accept” or press pound (or hash).</td>
</tr>
<tr>
<td>239</td>
<td>The envelope is turned off.</td>
<td>N/A</td>
</tr>
<tr>
<td>240</td>
<td>Please enter the application extension number.</td>
<td>Please say or enter the application extension number.</td>
</tr>
<tr>
<td>241</td>
<td>To select a day recording, press one.</td>
<td>To select a day recording, say “Day” or press one.</td>
</tr>
<tr>
<td>242</td>
<td>To select a night recording, press two.</td>
<td>To select a night recording, say “Night” or press two.</td>
</tr>
<tr>
<td>243</td>
<td>If this is the correct recording, press pound (or hash).</td>
<td>If this is the correct recording, say “Yes” or press pound (or hash).</td>
</tr>
<tr>
<td>244</td>
<td>Warning, no recording has been made.</td>
<td>N/A</td>
</tr>
<tr>
<td>245</td>
<td>Invalid application entered.</td>
<td>N/A</td>
</tr>
<tr>
<td>246</td>
<td>...is not a valid recording number.</td>
<td>N/A</td>
</tr>
<tr>
<td>247 - 252</td>
<td>Not used</td>
<td>Not used</td>
</tr>
<tr>
<td>253</td>
<td>This is a return receipt verifying the message sent has been heard.</td>
<td>N/A</td>
</tr>
<tr>
<td>254</td>
<td>I’m sorry. Private messages cannot be forwarded.</td>
<td>N/A</td>
</tr>
<tr>
<td>255</td>
<td>This is the message center calling for...</td>
<td>N/A</td>
</tr>
</tbody>
</table>
Table 54: Voice Prompts: Non-ASR and ASR (Continued)

<table>
<thead>
<tr>
<th>Prompt ID</th>
<th>Content (Non-ASR)</th>
<th>Content (ASR)</th>
</tr>
</thead>
<tbody>
<tr>
<td>256</td>
<td>If you are the correct person, press pound (or hash).</td>
<td>If you are the correct person, say “Yes” or press pound (or hash).</td>
</tr>
<tr>
<td>257</td>
<td>...One</td>
<td>N/A</td>
</tr>
<tr>
<td>258</td>
<td>...Two</td>
<td>N/A</td>
</tr>
<tr>
<td>259</td>
<td>...Three</td>
<td>N/A</td>
</tr>
<tr>
<td>260</td>
<td>...Four</td>
<td>N/A</td>
</tr>
<tr>
<td>261</td>
<td>...Five</td>
<td>N/A</td>
</tr>
<tr>
<td>262</td>
<td>...Six</td>
<td>N/A</td>
</tr>
<tr>
<td>263</td>
<td>...Seven</td>
<td>N/A</td>
</tr>
<tr>
<td>264</td>
<td>...Eight</td>
<td>N/A</td>
</tr>
<tr>
<td>265</td>
<td>...Nine</td>
<td>N/A</td>
</tr>
<tr>
<td>266</td>
<td>...Ten</td>
<td>N/A</td>
</tr>
<tr>
<td>267</td>
<td>...Eleven</td>
<td>N/A</td>
</tr>
<tr>
<td>268</td>
<td>...Twelve</td>
<td>N/A</td>
</tr>
<tr>
<td>269</td>
<td>...Thirteen</td>
<td>N/A</td>
</tr>
<tr>
<td>270</td>
<td>...Fourteen</td>
<td>N/A</td>
</tr>
<tr>
<td>271</td>
<td>...Fifteen</td>
<td>N/A</td>
</tr>
<tr>
<td>272</td>
<td>...Sixteen</td>
<td>N/A</td>
</tr>
<tr>
<td>273</td>
<td>...Seventeen</td>
<td>N/A</td>
</tr>
<tr>
<td>274</td>
<td>...Eighteen</td>
<td>N/A</td>
</tr>
<tr>
<td>275</td>
<td>...Nineteen</td>
<td>N/A</td>
</tr>
<tr>
<td>276</td>
<td>...Twenty</td>
<td>N/A</td>
</tr>
<tr>
<td>277</td>
<td>...Thirty</td>
<td>N/A</td>
</tr>
<tr>
<td>278</td>
<td>...Forty</td>
<td>N/A</td>
</tr>
<tr>
<td>279</td>
<td>...Fifty</td>
<td>N/A</td>
</tr>
<tr>
<td>280</td>
<td>...Sixty</td>
<td>N/A</td>
</tr>
<tr>
<td>281</td>
<td>...Seventy</td>
<td>N/A</td>
</tr>
<tr>
<td>282</td>
<td>...Eighty</td>
<td>N/A</td>
</tr>
<tr>
<td>283</td>
<td>...Ninety</td>
<td>N/A</td>
</tr>
<tr>
<td>284</td>
<td>...Hundred</td>
<td>N/A</td>
</tr>
<tr>
<td>285</td>
<td>This mailbox is currently full and cannot receive new messages.</td>
<td>N/A</td>
</tr>
<tr>
<td>286</td>
<td>Oh</td>
<td>N/A</td>
</tr>
<tr>
<td>287</td>
<td>...of which are priority.</td>
<td>N/A</td>
</tr>
<tr>
<td>288</td>
<td>Priority message received...</td>
<td>N/A</td>
</tr>
<tr>
<td>289</td>
<td>Otherwise, press star.</td>
<td>Otherwise, say “No” or press star.</td>
</tr>
</tbody>
</table>
### Table 54: Voice Prompts: Non-ASR and ASR (Continued)

<table>
<thead>
<tr>
<th>Prompt ID</th>
<th>Content (Non-ASR)</th>
<th>Content (ASR)</th>
</tr>
</thead>
<tbody>
<tr>
<td>290</td>
<td>When you are finished entering mailbox numbers, press pound (or hash).</td>
<td>N/A</td>
</tr>
<tr>
<td>291</td>
<td>Otherwise, press pound (or hash).</td>
<td>Otherwise, say “Send” or press pound (or hash).</td>
</tr>
<tr>
<td>292</td>
<td>For PM, press two.</td>
<td>For PM, say “PM” or press two.</td>
</tr>
<tr>
<td>293</td>
<td>For Sunday, press one.</td>
<td>For Sunday, say “Sunday” or press one.</td>
</tr>
<tr>
<td>294</td>
<td>For Monday, press two.</td>
<td>For Monday, say “Monday” or press two.</td>
</tr>
<tr>
<td>295</td>
<td>For Tuesday, press three.</td>
<td>For Tuesday, say “Tuesday” or press three.</td>
</tr>
<tr>
<td>296</td>
<td>For Wednesday, press four.</td>
<td>For Wednesday, say “Wednesday” or press four.</td>
</tr>
<tr>
<td>297</td>
<td>For Thursday, press five.</td>
<td>For Thursday, say “Thursday” or press five.</td>
</tr>
<tr>
<td>298</td>
<td>For Friday, press six.</td>
<td>For Friday, say “Friday” or press six.</td>
</tr>
<tr>
<td>299</td>
<td>For Saturday, press seven.</td>
<td>For Saturday, say “Saturday” or press seven.</td>
</tr>
<tr>
<td>300</td>
<td>For Monday through Friday service, press one.</td>
<td>For Monday through Friday service, say “Week-days” or press one.</td>
</tr>
<tr>
<td>301</td>
<td>For all the days of the week, press two.</td>
<td>For all the days of the week, say “Days” or press two.</td>
</tr>
<tr>
<td>302*</td>
<td>Thousand</td>
<td>N/A</td>
</tr>
<tr>
<td>303</td>
<td>The hard disk is...</td>
<td>N/A</td>
</tr>
<tr>
<td>304</td>
<td>...percent full.</td>
<td>N/A</td>
</tr>
<tr>
<td>305</td>
<td>I’m sorry. The system is full and cannot record any messages.</td>
<td>N/A</td>
</tr>
<tr>
<td>306*</td>
<td>Zero</td>
<td>N/A</td>
</tr>
<tr>
<td>307</td>
<td>pound (or hash)</td>
<td>N/A</td>
</tr>
<tr>
<td>308</td>
<td>star</td>
<td>N/A</td>
</tr>
<tr>
<td>309</td>
<td>Group list number...</td>
<td>N/A</td>
</tr>
<tr>
<td>310</td>
<td>Please speak louder.</td>
<td>N/A</td>
</tr>
<tr>
<td>311</td>
<td>You have recorded for the maximum time permitted.</td>
<td>N/A</td>
</tr>
<tr>
<td>312</td>
<td>Not used</td>
<td>Not used</td>
</tr>
<tr>
<td>313</td>
<td>Resuming playback.</td>
<td>N/A</td>
</tr>
<tr>
<td>314</td>
<td>You have one priority message.</td>
<td>N/A</td>
</tr>
<tr>
<td>315</td>
<td>Use two digits for the hour and two digits for the minutes.</td>
<td>N/A</td>
</tr>
<tr>
<td>316 - 317</td>
<td>Not used</td>
<td>Not used</td>
</tr>
<tr>
<td>318</td>
<td>You have chosen not to have a password.</td>
<td>N/A</td>
</tr>
<tr>
<td>319</td>
<td>I’m sorry. This mailbox is currently in use and cannot be accessed. Please try again later.</td>
<td>N/A</td>
</tr>
<tr>
<td>320</td>
<td>Not used</td>
<td>Not used</td>
</tr>
</tbody>
</table>
### Table 54: Voice Prompts: Non-ASR and ASR (Continued)

<table>
<thead>
<tr>
<th>Prompt ID</th>
<th>Content (Non-ASR)</th>
<th>Content (ASR)</th>
</tr>
</thead>
<tbody>
<tr>
<td>321</td>
<td>If you are calling from a rotary-type telephone, please stay on the line and your call will be answered.</td>
<td>N/A</td>
</tr>
<tr>
<td>322</td>
<td>You can not reply to a system sent message.</td>
<td>N/A</td>
</tr>
<tr>
<td>323 - 324</td>
<td>Not used</td>
<td>Not used</td>
</tr>
<tr>
<td>325</td>
<td>Please select the days of the week.</td>
<td>N/A</td>
</tr>
<tr>
<td>326</td>
<td>To select individual week days, press three.</td>
<td>To select individual week days, say “Day” or press three.</td>
</tr>
<tr>
<td>327</td>
<td>Service is set for Monday through Friday.</td>
<td>N/A</td>
</tr>
<tr>
<td>328</td>
<td>Service is set for each day of the week.</td>
<td>N/A</td>
</tr>
<tr>
<td>329</td>
<td>Recording canceled.</td>
<td>N/A</td>
</tr>
<tr>
<td>330</td>
<td>The system was busy and could not deliver message notification at the specified time.</td>
<td>N/A</td>
</tr>
<tr>
<td>331</td>
<td>The system was unable to deliver message notification due to a programming error.</td>
<td>N/A</td>
</tr>
<tr>
<td>332</td>
<td>Extension ID...</td>
<td>N/A</td>
</tr>
<tr>
<td>333</td>
<td>...the operator.</td>
<td>N/A</td>
</tr>
<tr>
<td>334</td>
<td>...the operator...</td>
<td>N/A</td>
</tr>
<tr>
<td>335</td>
<td>Remote Messaging will call...</td>
<td>N/A</td>
</tr>
<tr>
<td>336</td>
<td>To change a cascade level, press one.</td>
<td>To change a cascade level, say “Level” or press one.</td>
</tr>
<tr>
<td>337</td>
<td>To select a message notification category, press four.</td>
<td>To select a message notification category, say “Category” or press four.</td>
</tr>
<tr>
<td>338</td>
<td>Please enter the cascade level you wish to program.</td>
<td>Please enter the cascade level you wish to program. Say or enter one through nine using the keys on your telephone.</td>
</tr>
<tr>
<td>339</td>
<td>...with a pager number of...</td>
<td>N/A</td>
</tr>
<tr>
<td>340</td>
<td>To change the pager dial string, press three.</td>
<td>To change the pager dial string, say “Pager” or press three.</td>
</tr>
<tr>
<td>341</td>
<td>To call an internal extension, press one.</td>
<td>To call an internal extension, say “Internal” or press one.</td>
</tr>
<tr>
<td>342</td>
<td>To call an outside number, press two.</td>
<td>To call an outside number, say “Outside” or press two.</td>
</tr>
<tr>
<td>343</td>
<td>Not used</td>
<td>Not used</td>
</tr>
<tr>
<td>344</td>
<td>Please enter the digits for your digital pager and wait for further instructions.</td>
<td>Please say or enter the digits for your digital pager and wait for further instructions.</td>
</tr>
<tr>
<td>345</td>
<td>The new pager number is...</td>
<td>N/A</td>
</tr>
<tr>
<td>346</td>
<td>Mailbox Number</td>
<td>N/A</td>
</tr>
<tr>
<td>347</td>
<td>New Message Count</td>
<td>N/A</td>
</tr>
<tr>
<td>348</td>
<td>Pause</td>
<td>N/A</td>
</tr>
</tbody>
</table>
### Table 54: Voice Prompts: Non-ASR and ASR (Continued)

<table>
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<tr>
<th>Prompt ID</th>
<th>Content (Non-ASR)</th>
<th>Content (ASR)</th>
</tr>
</thead>
<tbody>
<tr>
<td>349</td>
<td>You must program a telephone number first.</td>
<td>N/A</td>
</tr>
<tr>
<td>350</td>
<td>You have chosen not to have a pager number.</td>
<td>N/A</td>
</tr>
<tr>
<td>351</td>
<td>Cascade Level...</td>
<td>N/A</td>
</tr>
<tr>
<td>352</td>
<td>You did not enter a valid document number.</td>
<td>N/A</td>
</tr>
<tr>
<td>353</td>
<td>This document is currently unavailable.</td>
<td>N/A</td>
</tr>
<tr>
<td>354</td>
<td>... is not a valid document number.</td>
<td>N/A</td>
</tr>
<tr>
<td>355</td>
<td>You have already selected that document.</td>
<td>N/A</td>
</tr>
<tr>
<td>356</td>
<td>To cancel this message, press star.</td>
<td>To cancel this message, say “Cancel” or press star.</td>
</tr>
<tr>
<td>357</td>
<td>You have selected the maximum number of documents allowed.</td>
<td>N/A</td>
</tr>
<tr>
<td>358</td>
<td>To send your fax, press pound (or hash).</td>
<td>To send your fax, say “Send” or press pound (or hash).</td>
</tr>
<tr>
<td>359</td>
<td>You have selected one document.</td>
<td>N/A</td>
</tr>
<tr>
<td>360</td>
<td>You haven’t selected any documents yet.</td>
<td>N/A</td>
</tr>
<tr>
<td>361</td>
<td>You may select one more document during this call.</td>
<td>N/A</td>
</tr>
<tr>
<td>362</td>
<td>You may select up to...</td>
<td>N/A</td>
</tr>
<tr>
<td>363</td>
<td>... more documents during this call.</td>
<td>N/A</td>
</tr>
<tr>
<td>364</td>
<td>Selections canceled.</td>
<td>N/A</td>
</tr>
<tr>
<td>365</td>
<td>The number that will appear on your fax cover sheet is...</td>
<td>N/A</td>
</tr>
<tr>
<td>366</td>
<td>You have chosen not to have your extension appear on your cover sheet.</td>
<td>N/A</td>
</tr>
<tr>
<td>367</td>
<td>If you would like to have your fax delivered in the future, press pound (or hash).</td>
<td>If you would like to have your fax delivered in the future, say “Delay” or press pound (or hash).</td>
</tr>
<tr>
<td>368</td>
<td>To re-enter the number, press three.</td>
<td>To re-enter the number, say “Re-enter” or press three.</td>
</tr>
<tr>
<td>369</td>
<td>If you wish to have your extension appear on your cover sheet, enter the extension followed by pound (or hash).</td>
<td>If you wish to have your extension appear on your cover sheet, say or enter the extension then say or press pound (or hash).</td>
</tr>
<tr>
<td>370</td>
<td>If you are calling from your fax machine, press one.</td>
<td>If you are calling from your fax machine, say “Yes” or press one.</td>
</tr>
<tr>
<td>371</td>
<td>Press the start button on your fax machine now.</td>
<td>N/A</td>
</tr>
<tr>
<td>372</td>
<td>All of our fax lines are currently busy.</td>
<td>N/A</td>
</tr>
<tr>
<td>373</td>
<td>Please try again later.</td>
<td>N/A</td>
</tr>
<tr>
<td>374</td>
<td>Your fax will be delivered to...</td>
<td>N/A</td>
</tr>
<tr>
<td>375</td>
<td>Your fax will be delivered shortly.</td>
<td>N/A</td>
</tr>
</tbody>
</table>
Table 54: Voice Prompts: Non-ASR and ASR (Continued)

<table>
<thead>
<tr>
<th>Prompt ID</th>
<th>Content (Non-ASR)</th>
<th>Content (ASR)</th>
</tr>
</thead>
<tbody>
<tr>
<td>376</td>
<td>I am unable to communicate with your fax machine. Please make sure it is operating properly and try again.</td>
<td>N/A</td>
</tr>
<tr>
<td>377</td>
<td><em>(For U.S. systems)</em> Please enter the 10 digit telephone number of your fax machine, including area code. <em>(For European systems)</em> Please enter the telephone number of your fax machine, including your national dialing code if required.</td>
<td><em>(For U.S. systems)</em> Please say or enter the 10 digit fax number, including area code. <em>(For European systems)</em> Please say or enter the telephone number of your fax machine, including your national dialing code if required.</td>
</tr>
<tr>
<td>378</td>
<td>The number you enter may be up to twenty digits in length.</td>
<td>N/A</td>
</tr>
<tr>
<td>379</td>
<td>Your fax is scheduled to be delivered at...</td>
<td>N/A</td>
</tr>
<tr>
<td>380</td>
<td>Please enter a document number.</td>
<td>Please say or enter the document number.</td>
</tr>
<tr>
<td>381</td>
<td>When you have finished importing documents, press star.</td>
<td>When you are done importing documents, say “Finished” or press star.</td>
</tr>
<tr>
<td>382</td>
<td>Document number...</td>
<td>N/A</td>
</tr>
<tr>
<td>383</td>
<td>...is currently in use.</td>
<td>N/A</td>
</tr>
<tr>
<td>384</td>
<td>...already exists.</td>
<td>N/A</td>
</tr>
<tr>
<td>385</td>
<td>To replace this document, press pound (or hash).</td>
<td>To replace this document, say “Replace” or press pound (or hash).</td>
</tr>
<tr>
<td>386</td>
<td>Document received.</td>
<td>N/A</td>
</tr>
<tr>
<td>387</td>
<td>Press pound (or hash) to continue.</td>
<td>Say “Continue” or press pound (or hash) to continue.</td>
</tr>
<tr>
<td>388</td>
<td>To import a fax document, press four.</td>
<td>To import a fax document, say “Import Fax” or press four.</td>
</tr>
<tr>
<td>389</td>
<td><em>(For U.S. systems)</em> If this is an international call, enter 011 followed by the country code and fax number. Then press pound. <em>(For European systems)</em> If this is an international call, enter 00 code followed by your country code and your fax number. Then press hash.</td>
<td><em>(For U.S. systems)</em> If this is an international call, say or enter 011 followed by the country code and fax number. Then say or press pound/. <em>(For European systems)</em> If this is an international call, say or enter 00 code followed by your country code and your fax number. Then press hash.</td>
</tr>
<tr>
<td>390</td>
<td>I’m sorry. I’m not allowed to dial that number.</td>
<td>N/A</td>
</tr>
<tr>
<td>391</td>
<td>To receive your documents, please hang up and call from your fax machine.</td>
<td>N/A</td>
</tr>
<tr>
<td>392</td>
<td>The system does not have enough disk space to import your document.</td>
<td>N/A</td>
</tr>
<tr>
<td>393</td>
<td>The system is unable to import your document because it will exceed the Maximum Fax Library Size.</td>
<td>N/A</td>
</tr>
<tr>
<td>394</td>
<td>For recording options, press three.</td>
<td>For recording options, say “Options” or press three.</td>
</tr>
<tr>
<td>395</td>
<td>There are...</td>
<td>N/A</td>
</tr>
<tr>
<td>396</td>
<td>... calls ahead of you.</td>
<td>N/A</td>
</tr>
</tbody>
</table>
Table 54: Voice Prompts: Non-ASR and ASR (Continued)

<table>
<thead>
<tr>
<th>Prompt ID</th>
<th>Content (Non-ASR)</th>
<th>Content (ASR)</th>
</tr>
</thead>
<tbody>
<tr>
<td>397</td>
<td>Your call should be answered within...</td>
<td>N/A</td>
</tr>
<tr>
<td>398</td>
<td>You have selected...</td>
<td>N/A</td>
</tr>
<tr>
<td>399</td>
<td>...documents.</td>
<td>N/A</td>
</tr>
<tr>
<td>400</td>
<td>Hello, you have reached the Inter-Tel voice processing system.</td>
<td>N/A</td>
</tr>
<tr>
<td>401</td>
<td>Hello, you’ve reached the Inter-Tel automated call handling system.</td>
<td>N/A</td>
</tr>
<tr>
<td>402</td>
<td>To change your message search order, press two.</td>
<td>To change your message search order, say “Order” or press two.</td>
</tr>
<tr>
<td>403</td>
<td>Your current message search order is first in first out.</td>
<td>N/A</td>
</tr>
<tr>
<td>404</td>
<td>Your current message search order is last in, first out.</td>
<td>N/A</td>
</tr>
<tr>
<td>405</td>
<td>To search for messages first in first out, press one.</td>
<td>To search for messages first in first out, say “First In” or press one.</td>
</tr>
<tr>
<td>406</td>
<td>To search for messages last in first out, press two.</td>
<td>To search for messages last in first out, say “Last In” or press two.</td>
</tr>
<tr>
<td>407</td>
<td>To change the search order for new messages, press one.</td>
<td>To change the search order for new messages, say “New” or press one.</td>
</tr>
<tr>
<td>408</td>
<td>To change the search order for saved messages, press two.</td>
<td>To change the search order for saved messages, say “Saved” or press two.</td>
</tr>
<tr>
<td>409</td>
<td>You have received a new message</td>
<td>N/A</td>
</tr>
<tr>
<td>410</td>
<td>You have received a new voice mail</td>
<td>N/A</td>
</tr>
<tr>
<td>411</td>
<td>You have received a new e-mail</td>
<td>N/A</td>
</tr>
<tr>
<td>412</td>
<td>You have received a new fax</td>
<td>N/A</td>
</tr>
<tr>
<td>413</td>
<td>You have received...</td>
<td>N/A</td>
</tr>
<tr>
<td>414 - 499</td>
<td>Not used</td>
<td>Not used</td>
</tr>
<tr>
<td>500</td>
<td>If you know your party’s extension number, enter it now.</td>
<td>If you know your party’s extension, please say or enter it now.</td>
</tr>
<tr>
<td>501</td>
<td>To leave a message, press one.</td>
<td>To leave a message, say “Leave Message” or press one.</td>
</tr>
<tr>
<td>502</td>
<td>To try another extension, press two.</td>
<td>To try another extension, say “Extension” or press two.</td>
</tr>
<tr>
<td>503</td>
<td>Please hold while your call is being transferred to...</td>
<td>N/A</td>
</tr>
<tr>
<td>504</td>
<td>Not used</td>
<td>Not used</td>
</tr>
<tr>
<td>505</td>
<td>To return to the menu, press two.</td>
<td>To return to the menu, say “Menu” or press two.</td>
</tr>
<tr>
<td>506</td>
<td>Your fax is scheduled to be delivered on...</td>
<td>N/A</td>
</tr>
<tr>
<td>507</td>
<td>The system already contains the maximum number of fax documents allowed.</td>
<td>N/A</td>
</tr>
</tbody>
</table>
Table 54: Voice Prompts: Non-ASR and ASR (Continued)

<table>
<thead>
<tr>
<th>Prompt ID</th>
<th>Content (Non-ASR)</th>
<th>Content (ASR)</th>
</tr>
</thead>
<tbody>
<tr>
<td>508</td>
<td>The system does not have enough disk space to import any more documents.</td>
<td>N/A</td>
</tr>
<tr>
<td>509</td>
<td>The system is currently at or above the Maximum Fax Library Size.</td>
<td>N/A</td>
</tr>
<tr>
<td>510</td>
<td>Levels one through nine are currently disabled.</td>
<td>N/A</td>
</tr>
<tr>
<td>511</td>
<td>Your call will be answered in the order it was received.</td>
<td>N/A</td>
</tr>
<tr>
<td>512</td>
<td>There is one call ahead of you.</td>
<td>N/A</td>
</tr>
<tr>
<td>513</td>
<td>To change your transfer method, press five.</td>
<td>To change your transfer method, say “Transfer” or press five.</td>
</tr>
<tr>
<td>514</td>
<td>Your current transfer method is unannounced.</td>
<td>N/A</td>
</tr>
<tr>
<td>515</td>
<td>Your current transfer method is screened.</td>
<td>N/A</td>
</tr>
<tr>
<td>516</td>
<td>Your current transfer method is announce only.</td>
<td>N/A</td>
</tr>
<tr>
<td>517</td>
<td>To select unannounced transfers, press one.</td>
<td>To select unannounced transfers, say “Unannounced” or press one.</td>
</tr>
<tr>
<td>518</td>
<td>For screened transfers, press two.</td>
<td>For screened transfers, say “Screened” or press two.</td>
</tr>
<tr>
<td>519</td>
<td>For announce only, press three.</td>
<td>For announce only, say “Announce” or press three.</td>
</tr>
<tr>
<td>520</td>
<td>Who should I say is calling?</td>
<td>N/A</td>
</tr>
<tr>
<td>521</td>
<td>To replay the announcement, press one.</td>
<td>To replay the announcement, say “Replay” or press one.</td>
</tr>
<tr>
<td>522</td>
<td>To send this call to Voice Mail, press two.</td>
<td>To send this call to voice mail, say “Voice Mail” or press two.</td>
</tr>
<tr>
<td>523</td>
<td>To forward this call to another extension, press three.</td>
<td>To forward this call to another extension, say “Forward” or press three.</td>
</tr>
<tr>
<td>524</td>
<td>To accept this call, press pound (or hash).</td>
<td>To accept this call, say “Accept” or press pound (or hash).</td>
</tr>
<tr>
<td>525</td>
<td>To refuse this call, press star.</td>
<td>To refuse this call, say “Refuse” or press star.</td>
</tr>
<tr>
<td>526</td>
<td>You have a call from...</td>
<td>N/A</td>
</tr>
<tr>
<td>527</td>
<td>To change the time and date, press five.</td>
<td>N/A</td>
</tr>
<tr>
<td>528</td>
<td>The current time is...</td>
<td>N/A</td>
</tr>
<tr>
<td>529</td>
<td>To accept the current time, press pound (or hash).</td>
<td>N/A</td>
</tr>
<tr>
<td>530</td>
<td>To enter a new time, use two digits for the hour and two digits for the minutes.</td>
<td>N/A</td>
</tr>
<tr>
<td>531</td>
<td>The current date is...</td>
<td>N/A</td>
</tr>
<tr>
<td>532</td>
<td>To accept the current date, press pound (or hash).</td>
<td>N/A</td>
</tr>
<tr>
<td>533</td>
<td>To enter a new date, use two digits for the month, two digits for the day, and two digits for the year.</td>
<td>N/A</td>
</tr>
<tr>
<td>534</td>
<td>Invalid date.</td>
<td>N/A</td>
</tr>
</tbody>
</table>
### Table 54: Voice Prompts: Non-ASR and ASR (Continued)

<table>
<thead>
<tr>
<th>Prompt ID</th>
<th>Content (Non-ASR)</th>
<th>Content (ASR)</th>
</tr>
</thead>
<tbody>
<tr>
<td>535</td>
<td>Please enter the extension number. Please say or enter the extension number.</td>
<td></td>
</tr>
<tr>
<td>536 - 537</td>
<td>Not used</td>
<td>Not used</td>
</tr>
<tr>
<td>538</td>
<td>For system prompt replacement or reinstatement, press six.</td>
<td>For system prompt replacement or reinstatement, say “Replacement” or press six.</td>
</tr>
<tr>
<td>539</td>
<td>Please enter the number of the system prompt.</td>
<td>Please say or enter the number of the system prompt.</td>
</tr>
<tr>
<td>540</td>
<td>The system prompt has been replaced by the following custom prompt...</td>
<td>N/A</td>
</tr>
<tr>
<td>541</td>
<td>The original system prompt is...</td>
<td>N/A</td>
</tr>
<tr>
<td>542</td>
<td>To replace this system prompt with a custom prompt, press three.</td>
<td>To replace this system prompt with a custom prompt, say “Custom” or press three.</td>
</tr>
<tr>
<td>543</td>
<td>...is not a valid system prompt number.</td>
<td>N/A</td>
</tr>
<tr>
<td>544</td>
<td>To delete the custom prompt and reinstate the system prompt, press nine.</td>
<td>To delete the custom prompt and reinstate the system prompt, say “System” or press nine.</td>
</tr>
<tr>
<td>545</td>
<td>To replace this custom prompt with a new custom prompt, press three.</td>
<td>To replace this custom prompt with a new custom prompt, say “New Prompt” or press three.</td>
</tr>
<tr>
<td>546</td>
<td>You have no deleted messages.</td>
<td>N/A</td>
</tr>
<tr>
<td>547</td>
<td>You have one deleted message.</td>
<td>N/A</td>
</tr>
<tr>
<td>548</td>
<td>...deleted messages.</td>
<td>N/A</td>
</tr>
<tr>
<td>549</td>
<td>To recover deleted messages, press two.</td>
<td>To recover deleted messages, say “Recover” or press two.</td>
</tr>
<tr>
<td>550</td>
<td>To recover this message, press seven.</td>
<td>To recover this message, say “Recover” or press seven.</td>
</tr>
<tr>
<td>551</td>
<td>Message recovered.</td>
<td>N/A</td>
</tr>
<tr>
<td>552</td>
<td>To purge this message, press nine.</td>
<td>To purge this message, say “Purge” or press nine.</td>
</tr>
<tr>
<td>553</td>
<td>Message purged.</td>
<td>N/A</td>
</tr>
<tr>
<td>554</td>
<td>To listen to deleted messages, press one.</td>
<td>To listen to deleted messages, say “Listen” or press one.</td>
</tr>
<tr>
<td>555</td>
<td>To recover all messages, press two.</td>
<td>To recover all messages, say “Recover” or press two.</td>
</tr>
<tr>
<td>556</td>
<td>To purge all messages, press three.</td>
<td>To purge all messages, say “Purge” or press three.</td>
</tr>
<tr>
<td>557</td>
<td>...messages recovered.</td>
<td>N/A</td>
</tr>
<tr>
<td>558</td>
<td>...messages purged.</td>
<td>N/A</td>
</tr>
<tr>
<td>559</td>
<td>...message recovered.</td>
<td>N/A</td>
</tr>
<tr>
<td>560</td>
<td>...message purged.</td>
<td>N/A</td>
</tr>
<tr>
<td>561</td>
<td>For message options, press five.</td>
<td>For message options, say “Options” or press five.</td>
</tr>
<tr>
<td>562</td>
<td>For more options, press nine.</td>
<td>For more options, say “More” or press nine.</td>
</tr>
<tr>
<td>563</td>
<td>To change the fax destination number, press one.</td>
<td>To change the fax destination number, say “Number” or press one.</td>
</tr>
</tbody>
</table>
### Table 54: Voice Prompts: Non-ASR and ASR (Continued)

<table>
<thead>
<tr>
<th>Prompt ID</th>
<th>Content (Non-ASR)</th>
<th>Content (ASR)</th>
</tr>
</thead>
<tbody>
<tr>
<td>564</td>
<td>Please enter the extension number of your fax destination, followed by the pound (or hash) key.</td>
<td>Please say or enter the extension number of your fax. Then say or press pound (or hash).</td>
</tr>
<tr>
<td>565</td>
<td>Your fax destination is...</td>
<td>N/A</td>
</tr>
<tr>
<td>566</td>
<td>You have chosen not to have a fax destination.</td>
<td>N/A</td>
</tr>
<tr>
<td>567</td>
<td>Invalid extension.</td>
<td>N/A</td>
</tr>
<tr>
<td>568</td>
<td>Fax destination saved.</td>
<td>N/A</td>
</tr>
<tr>
<td>569</td>
<td>Fax destination erased.</td>
<td>N/A</td>
</tr>
<tr>
<td>570</td>
<td>To cancel unheard sent voice mail messages, press one.</td>
<td>To cancel unheard sent voice mail messages, say “Unheard” or press one.</td>
</tr>
<tr>
<td>571</td>
<td>All sent messages have been heard.</td>
<td>N/A</td>
</tr>
<tr>
<td>572</td>
<td>Enter the destination mailbox.</td>
<td>Say or enter the destination mailbox.</td>
</tr>
<tr>
<td>573</td>
<td>This mailbox has one unheard message.</td>
<td>N/A</td>
</tr>
<tr>
<td>574</td>
<td>…unheard messages.</td>
<td>N/A</td>
</tr>
<tr>
<td>575</td>
<td>To append to the message, press two.</td>
<td>To append the message, say “Append” or press two.</td>
</tr>
<tr>
<td>576</td>
<td>To leave a Voice Mail message, press one.</td>
<td>To leave a Voice Mail message, say “Voice Mail” or press one.</td>
</tr>
<tr>
<td>577</td>
<td>To make a return call, press two.</td>
<td>To make a return call, say “Call Back” or press two.</td>
</tr>
<tr>
<td>578</td>
<td>Please hold while your call is being transferred to an outside number.</td>
<td>N/A</td>
</tr>
<tr>
<td>579</td>
<td>Enter the number of the remote mailbox.</td>
<td>Say or enter the number of the remote mailbox.</td>
</tr>
<tr>
<td>580</td>
<td>… followed by pound (or hash).</td>
<td>N/A</td>
</tr>
<tr>
<td>581</td>
<td>…has just listened to the original message.</td>
<td>N/A</td>
</tr>
<tr>
<td>582</td>
<td>This mailbox has...</td>
<td>N/A</td>
</tr>
<tr>
<td>583 - 586</td>
<td>Not used</td>
<td>Not used</td>
</tr>
<tr>
<td>587</td>
<td>To hear the recipients, press three.</td>
<td>To hear the recipients, say “Hear” or press three.</td>
</tr>
<tr>
<td>588 - 591</td>
<td>Not used</td>
<td>Not used</td>
</tr>
<tr>
<td>592</td>
<td>I’m sorry. I’m not allowed to dial that number.</td>
<td>N/A</td>
</tr>
<tr>
<td>593</td>
<td>Message received from an outside number with no caller ID information.</td>
<td>N/A</td>
</tr>
<tr>
<td>594</td>
<td>I’m sorry. I do not have valid Caller ID information.</td>
<td>N/A</td>
</tr>
<tr>
<td>595</td>
<td>…the network…</td>
<td>N/A</td>
</tr>
<tr>
<td>596</td>
<td>Your call is being handled by the Inter-Tel voice processing system.</td>
<td>N/A</td>
</tr>
<tr>
<td>597</td>
<td>You have no new e-mails.</td>
<td>N/A</td>
</tr>
<tr>
<td>598</td>
<td>… new e-mails</td>
<td>N/A</td>
</tr>
<tr>
<td>599</td>
<td>you have one new e-mail.</td>
<td>N/A</td>
</tr>
<tr>
<td>600</td>
<td>…saved e-mails.</td>
<td>N/A</td>
</tr>
</tbody>
</table>
### Table 54: Voice Prompts: Non-ASR and ASR (Continued)

<table>
<thead>
<tr>
<th>Prompt ID</th>
<th>Content (Non-ASR)</th>
<th>Content (ASR)</th>
</tr>
</thead>
<tbody>
<tr>
<td>601</td>
<td>You have one saved e-mail.</td>
<td>N/A</td>
</tr>
<tr>
<td>602</td>
<td>You have no new faxes.</td>
<td>N/A</td>
</tr>
<tr>
<td>603</td>
<td>... new faxes.</td>
<td>N/A</td>
</tr>
<tr>
<td>604</td>
<td>You have one new fax.</td>
<td>N/A</td>
</tr>
<tr>
<td>605</td>
<td>...saved faxes.</td>
<td>N/A</td>
</tr>
<tr>
<td>606</td>
<td>You have one saved fax.</td>
<td>N/A</td>
</tr>
<tr>
<td>607</td>
<td>The system has not been properly configured.</td>
<td>N/A</td>
</tr>
<tr>
<td>608</td>
<td>To listen to voice mails, press one.</td>
<td>To listen to voice mails, say “Voice Mail” or press one.</td>
</tr>
<tr>
<td>609</td>
<td>To listen to e-mails, press three.</td>
<td>To listen to e-mails, say “E-mail” or press three.</td>
</tr>
<tr>
<td>610</td>
<td>To listen to faxes, press six.</td>
<td>To listen to Fax messages, say “Facsimile” or press six.</td>
</tr>
<tr>
<td>611</td>
<td>You have no new messages.</td>
<td>N/A</td>
</tr>
<tr>
<td>612</td>
<td>with the subject...</td>
<td>N/A</td>
</tr>
<tr>
<td>613</td>
<td>I’m sorry, you cannot delete this e-mail.</td>
<td>N/A</td>
</tr>
<tr>
<td>614</td>
<td>To reply to all recipients, press one.</td>
<td>To reply to all recipients, say “Everyone” or press one.</td>
</tr>
<tr>
<td>615</td>
<td>To reply to just the sender, press two.</td>
<td>To reply to just the sender, say “Sender” or press two.</td>
</tr>
<tr>
<td>616</td>
<td>I’m sorry, this message was received from an unknown source.</td>
<td>N/A</td>
</tr>
<tr>
<td>617</td>
<td>Please enter the digits for the fax destination and wait for further instructions.</td>
<td>N/A</td>
</tr>
<tr>
<td>618</td>
<td>The system could not access your e-mail account with the known password.</td>
<td>N/A</td>
</tr>
<tr>
<td>619</td>
<td>...totalling</td>
<td>N/A</td>
</tr>
<tr>
<td>620</td>
<td>...pages</td>
<td>N/A</td>
</tr>
<tr>
<td>621</td>
<td>I’m sorry, you cannot delete this fax.</td>
<td>N/A</td>
</tr>
<tr>
<td>622</td>
<td>For voice mails, press one.</td>
<td>For voice mails, say “Voice Mail” or press one.</td>
</tr>
<tr>
<td>623</td>
<td>For e-mails, press two.</td>
<td>For e-mails, say “E-mail” or press two.</td>
</tr>
<tr>
<td>624</td>
<td>For faxes, press three.</td>
<td>For faxes, say “Facsimile” or press three.</td>
</tr>
<tr>
<td>625 - 629</td>
<td>Not used</td>
<td>Not used</td>
</tr>
<tr>
<td>630</td>
<td>voice mail</td>
<td>N/A</td>
</tr>
<tr>
<td>631</td>
<td>e-mail</td>
<td>N/A</td>
</tr>
<tr>
<td>632</td>
<td>Fax</td>
<td>N/A</td>
</tr>
<tr>
<td>633 - 644</td>
<td>Not used</td>
<td>Not used</td>
</tr>
<tr>
<td>645</td>
<td>...flagged...</td>
<td>N/A</td>
</tr>
<tr>
<td>646</td>
<td>...saved...</td>
<td>N/A</td>
</tr>
</tbody>
</table>
Table 54: Voice Prompts: Non-ASR and ASR (Continued)

<table>
<thead>
<tr>
<th>Prompt ID</th>
<th>Content (Non-ASR)</th>
<th>Content (ASR)</th>
</tr>
</thead>
<tbody>
<tr>
<td>647</td>
<td>...answered...</td>
<td>N/A</td>
</tr>
<tr>
<td>648</td>
<td>...recent...</td>
<td>N/A</td>
</tr>
<tr>
<td>649</td>
<td>...deleted...</td>
<td>N/A</td>
</tr>
<tr>
<td>650</td>
<td>...one page...</td>
<td>N/A</td>
</tr>
<tr>
<td>651</td>
<td>...voice mails...</td>
<td>N/A</td>
</tr>
<tr>
<td>652</td>
<td>...e-mails...</td>
<td>N/A</td>
</tr>
<tr>
<td>653</td>
<td>...faxes...</td>
<td>N/A</td>
</tr>
<tr>
<td>654</td>
<td>This message is marked...</td>
<td>N/A</td>
</tr>
<tr>
<td>655</td>
<td>...flagged...</td>
<td>N/A</td>
</tr>
<tr>
<td>656</td>
<td>...saved...</td>
<td>N/A</td>
</tr>
<tr>
<td>657</td>
<td>...answered...</td>
<td>N/A</td>
</tr>
<tr>
<td>658</td>
<td>...recent...</td>
<td>N/A</td>
</tr>
<tr>
<td>659</td>
<td>...deleted...</td>
<td>N/A</td>
</tr>
<tr>
<td>660</td>
<td>...with attachment...</td>
<td>N/A</td>
</tr>
<tr>
<td>661</td>
<td>For your voice mail envelope, press one.</td>
<td>For your voice mail envelope, say “Voice Mail” or press one.</td>
</tr>
<tr>
<td>662</td>
<td>For your e-mail envelope, press two.</td>
<td>For your e-mail envelope, say “E-mail” or press two.</td>
</tr>
<tr>
<td>663</td>
<td>For your fax envelope, press three.</td>
<td>For your fax envelope, say “Facsimile” or press three.</td>
</tr>
<tr>
<td>664</td>
<td>For message filters, press six.</td>
<td>For message filters, say “Filters” or press six.</td>
</tr>
<tr>
<td>665</td>
<td>For the message subject, press three.</td>
<td>For the message subject, say “Subject” or press three.</td>
</tr>
<tr>
<td>666</td>
<td>...message subject...</td>
<td>N/A</td>
</tr>
<tr>
<td>667</td>
<td>...message filters...</td>
<td>N/A</td>
</tr>
<tr>
<td>668</td>
<td>For the number of pages, press three.</td>
<td>For the number of pages, say “Pages” or press three.</td>
</tr>
<tr>
<td>669</td>
<td>You cannot turn all fax envelope options off.</td>
<td>N/A</td>
</tr>
<tr>
<td>670</td>
<td>...number of pages...</td>
<td>N/A</td>
</tr>
<tr>
<td>671</td>
<td>You have no new voice mails.</td>
<td>N/A</td>
</tr>
<tr>
<td>672</td>
<td>Fax scheduled for delivery.</td>
<td>N/A</td>
</tr>
<tr>
<td>673</td>
<td>...new voice mails.</td>
<td>N/A</td>
</tr>
<tr>
<td>674</td>
<td>You have one priority voice mail.</td>
<td>N/A</td>
</tr>
<tr>
<td>675</td>
<td>You have one new voice mail</td>
<td>N/A</td>
</tr>
<tr>
<td>676</td>
<td>...saved voice mails...</td>
<td>N/A</td>
</tr>
<tr>
<td>677</td>
<td>You have one saved voice mail.</td>
<td>N/A</td>
</tr>
<tr>
<td>678</td>
<td>To listen to messages, press one.</td>
<td>To listen to messages, say “Messages” or press one.</td>
</tr>
</tbody>
</table>
### Table 54: Voice Prompts: Non-ASR and ASR (Continued)

<table>
<thead>
<tr>
<th>Prompt ID</th>
<th>Content (Non-ASR)</th>
<th>Content (ASR)</th>
</tr>
</thead>
<tbody>
<tr>
<td>679</td>
<td>The system is momentarily unable to process your request.</td>
<td>The system is momentarily unable to process your request.</td>
</tr>
<tr>
<td>680</td>
<td>To wait for resources to become available, press one.</td>
<td>To wait for resources to become available, say “Wait” or press one.</td>
</tr>
<tr>
<td>681</td>
<td>To cancel the request and return to a previous menu, press star.</td>
<td>To cancel the request and return to a previous menu, say “Cancel” or press star.</td>
</tr>
<tr>
<td>682</td>
<td>Not used</td>
<td>Not used</td>
</tr>
<tr>
<td>683</td>
<td>You have one unheard voice mail.</td>
<td>N/A</td>
</tr>
<tr>
<td>684</td>
<td>You have one heard voice mail.</td>
<td>N/A</td>
</tr>
<tr>
<td>685</td>
<td>… unheard voice mail…</td>
<td>N/A</td>
</tr>
<tr>
<td>686</td>
<td>… heard voice mail…</td>
<td>N/A</td>
</tr>
<tr>
<td>687</td>
<td>You have no unheard voice mails.</td>
<td>N/A</td>
</tr>
<tr>
<td>688 - 692</td>
<td>Not used</td>
<td>Not used</td>
</tr>
<tr>
<td>693</td>
<td>You have one priority e-mail</td>
<td>N/A</td>
</tr>
<tr>
<td>694 - 699</td>
<td>Not used</td>
<td>Not used</td>
</tr>
<tr>
<td>700</td>
<td>N/A</td>
<td>Please spell the first few letters of where you would like to search in the directory.</td>
</tr>
<tr>
<td>701</td>
<td>...hours...</td>
<td>N/A</td>
</tr>
<tr>
<td>702</td>
<td>...hour...</td>
<td>N/A</td>
</tr>
<tr>
<td>703</td>
<td>&lt;10 second pause&gt;</td>
<td>N/A</td>
</tr>
<tr>
<td>704</td>
<td>&lt;five second pause&gt;</td>
<td>N/A</td>
</tr>
<tr>
<td>705*</td>
<td>&lt;one second pause&gt;</td>
<td>N/A</td>
</tr>
<tr>
<td>706</td>
<td>Fax canceled</td>
<td>N/A</td>
</tr>
<tr>
<td>707</td>
<td>phone number…</td>
<td>N/A</td>
</tr>
<tr>
<td>708</td>
<td>Not used</td>
<td>Not used</td>
</tr>
<tr>
<td>709</td>
<td>Your results match more than…</td>
<td>Your results match more than…</td>
</tr>
<tr>
<td>710</td>
<td>…entries.</td>
<td>…entries.</td>
</tr>
<tr>
<td>711</td>
<td>If you would like to browse these results, press one. To filter your search, press two.</td>
<td>If you would like to browse these results, please say “Yes” or press one. To filter your search, say “No” or press two.</td>
</tr>
<tr>
<td>712</td>
<td>If the result is the first name, press one. Otherwise, please tell me the first name by pressing the keys on your telephone.</td>
<td>If that name is the first name, say “Yes” or press one. Otherwise, please tell me the first name by saying, spelling, or pressing the keys on your telephone.</td>
</tr>
<tr>
<td>713</td>
<td>If the result is the last name, press one. Otherwise, please tell me the last name by pressing the keys on your telephone.</td>
<td>If that name is the last name, say “Yes” or press one. Otherwise, please tell me the last name by saying, spelling, or pressing the keys on your telephone.</td>
</tr>
<tr>
<td>715 - 716</td>
<td>Not used</td>
<td>Not used</td>
</tr>
<tr>
<td>717</td>
<td>…one saved voice mail</td>
<td>N/A</td>
</tr>
</tbody>
</table>
### Table 54: Voice Prompts: Non-ASR and ASR (Continued)

<table>
<thead>
<tr>
<th>Prompt ID</th>
<th>Content (Non-ASR)</th>
<th>Content (ASR)</th>
</tr>
</thead>
<tbody>
<tr>
<td>718</td>
<td>…one saved e-mail</td>
<td>N/A</td>
</tr>
<tr>
<td>719</td>
<td>…one saved fax</td>
<td>N/A</td>
</tr>
<tr>
<td>720</td>
<td>…one saved message</td>
<td>N/A</td>
</tr>
<tr>
<td>721</td>
<td>…unheard voice mails</td>
<td>N/A</td>
</tr>
<tr>
<td>722</td>
<td>…heard voice mails</td>
<td>N/A</td>
</tr>
<tr>
<td>723 - 726</td>
<td>Not used</td>
<td>Not used</td>
</tr>
<tr>
<td>727</td>
<td>…one heard voice mail</td>
<td>N/A</td>
</tr>
<tr>
<td>728 - 730</td>
<td>Not used</td>
<td>Not used</td>
</tr>
<tr>
<td>731</td>
<td>I’m sorry, I didn’t understand your response.</td>
<td>N/A</td>
</tr>
<tr>
<td>732</td>
<td>The system prompt you have selected has two formats.</td>
<td>N/A</td>
</tr>
<tr>
<td>733</td>
<td>To select the standard format, press one.</td>
<td>To select the standard format, say “Standard” or press one.</td>
</tr>
<tr>
<td>734</td>
<td>To select the speech-recognition format, press two.</td>
<td>To select the speech-recognition format, say “Speech Recognition” or press two.</td>
</tr>
<tr>
<td>735</td>
<td>To replace the standard format, press one.</td>
<td>To replace the standard format, say “Standard” or press one.</td>
</tr>
<tr>
<td>736</td>
<td>To replace the speech-recognition format, press one.</td>
<td>To replace the speech-recognition format, say “Speech Recognition” or press one.</td>
</tr>
<tr>
<td>737 - 738</td>
<td>Not used</td>
<td>Not used</td>
</tr>
<tr>
<td>739</td>
<td>Speech recognition has been disabled for this call.</td>
<td>N/A</td>
</tr>
<tr>
<td>740</td>
<td>Not used</td>
<td>Not used</td>
</tr>
<tr>
<td>741</td>
<td>… messages…</td>
<td>N/A</td>
</tr>
<tr>
<td>742</td>
<td>…seen...</td>
<td>N/A</td>
</tr>
<tr>
<td>743 - 744</td>
<td>Not used</td>
<td>Not used</td>
</tr>
<tr>
<td>745</td>
<td>You have no saved voice mails.</td>
<td>N/A</td>
</tr>
<tr>
<td>746</td>
<td>Not used</td>
<td>Not used</td>
</tr>
<tr>
<td>747</td>
<td>You have no saved e-mails.</td>
<td>N/A</td>
</tr>
<tr>
<td>748</td>
<td>Not used</td>
<td>Not used</td>
</tr>
<tr>
<td>749</td>
<td>You have no saved faxes.</td>
<td>N/A</td>
</tr>
<tr>
<td>750</td>
<td>Not used</td>
<td>Not used</td>
</tr>
<tr>
<td>751</td>
<td>voice mail received...</td>
<td>N/A</td>
</tr>
<tr>
<td>752</td>
<td>E-mail received...</td>
<td>N/A</td>
</tr>
<tr>
<td>753</td>
<td>Fax received...</td>
<td>N/A</td>
</tr>
<tr>
<td>754</td>
<td>Priority voice mail received...</td>
<td>N/A</td>
</tr>
<tr>
<td>755</td>
<td>Priority e-mail received...</td>
<td>N/A</td>
</tr>
<tr>
<td>756</td>
<td>Not used</td>
<td>Not used</td>
</tr>
</tbody>
</table>
### Table 54: Voice Prompts: Non-ASR and ASR (Continued)

<table>
<thead>
<tr>
<th>Prompt ID</th>
<th>Content (Non-ASR)</th>
<th>Content (ASR)</th>
</tr>
</thead>
<tbody>
<tr>
<td>757</td>
<td>and seen</td>
<td>N/A</td>
</tr>
<tr>
<td>758</td>
<td>and recent</td>
<td>N/A</td>
</tr>
<tr>
<td>759</td>
<td>and answered</td>
<td>N/A</td>
</tr>
<tr>
<td>760</td>
<td>and deleted</td>
<td>N/A</td>
</tr>
<tr>
<td>761</td>
<td>Not used</td>
<td>Not used</td>
</tr>
<tr>
<td>762*</td>
<td>...Million...</td>
<td>N/A</td>
</tr>
<tr>
<td>763*</td>
<td>...Billion...</td>
<td>N/A</td>
</tr>
<tr>
<td>764*</td>
<td>...Trillion...</td>
<td>N/A</td>
</tr>
<tr>
<td>765*</td>
<td>...Minus...</td>
<td>N/A</td>
</tr>
<tr>
<td>766*</td>
<td>(For U.S. systems) ...Dollar...</td>
<td>N/A</td>
</tr>
<tr>
<td></td>
<td>(For European systems)...Pound...</td>
<td></td>
</tr>
<tr>
<td>767*</td>
<td>(For U.S. systems) ...Dollars...</td>
<td>N/A</td>
</tr>
<tr>
<td></td>
<td>(For European systems)...Pounds...</td>
<td></td>
</tr>
<tr>
<td>768*</td>
<td>(For U.S. systems) ...Cent...</td>
<td>N/A</td>
</tr>
<tr>
<td></td>
<td>(For European systems)...Pence...</td>
<td></td>
</tr>
<tr>
<td>769*</td>
<td>(For U.S. systems) ...Cents...</td>
<td>N/A</td>
</tr>
<tr>
<td></td>
<td>(For European systems)...Pence...</td>
<td></td>
</tr>
<tr>
<td>770</td>
<td>&lt;Pause&gt;</td>
<td>N/A</td>
</tr>
<tr>
<td>771*</td>
<td>O’clock</td>
<td>N/A</td>
</tr>
<tr>
<td>772*</td>
<td>...Point...</td>
<td>N/A</td>
</tr>
<tr>
<td>773</td>
<td>I was unable to connect to your voice mail server.</td>
<td>N/A</td>
</tr>
<tr>
<td>774</td>
<td>I was unable to connect to your e-mail server.</td>
<td>N/A</td>
</tr>
<tr>
<td>775</td>
<td>I was unable to connect to your fax server.</td>
<td>N/A</td>
</tr>
<tr>
<td>776</td>
<td>If you have changed your password, press one.</td>
<td>If you have changed your password, say “Password” or press one.</td>
</tr>
<tr>
<td>777</td>
<td>To continue without these messages, press pound.</td>
<td>To continue without these messages, say “Continue” or press pound.</td>
</tr>
<tr>
<td>778*</td>
<td>dash</td>
<td>N/A</td>
</tr>
<tr>
<td>779*</td>
<td>ampersand</td>
<td>N/A</td>
</tr>
<tr>
<td>780*</td>
<td>at</td>
<td>N/A</td>
</tr>
<tr>
<td>781*</td>
<td>left parenthesis</td>
<td>N/A</td>
</tr>
<tr>
<td>782*</td>
<td>right parenthesis</td>
<td>N/A</td>
</tr>
<tr>
<td>783*</td>
<td>underscore</td>
<td>N/A</td>
</tr>
<tr>
<td>784*</td>
<td>space</td>
<td>N/A</td>
</tr>
<tr>
<td>785*</td>
<td>Tilde</td>
<td>N/A</td>
</tr>
<tr>
<td>786*</td>
<td>a</td>
<td>N/A</td>
</tr>
<tr>
<td>787*</td>
<td>b</td>
<td>N/A</td>
</tr>
<tr>
<td>788*</td>
<td>c</td>
<td>N/A</td>
</tr>
</tbody>
</table>
**Table 54: Voice Prompts: Non-ASR and ASR (Continued)**

<table>
<thead>
<tr>
<th>Prompt ID</th>
<th>Content (Non-ASR)</th>
<th>Content (ASR)</th>
</tr>
</thead>
<tbody>
<tr>
<td>789*</td>
<td>apostrophe</td>
<td>N/A</td>
</tr>
<tr>
<td>790*</td>
<td>lower A</td>
<td>N/A</td>
</tr>
<tr>
<td>791*</td>
<td>lower B</td>
<td>N/A</td>
</tr>
<tr>
<td>792*</td>
<td>lower C</td>
<td>N/A</td>
</tr>
<tr>
<td>793*</td>
<td>d</td>
<td>N/A</td>
</tr>
<tr>
<td>794*</td>
<td>e</td>
<td>N/A</td>
</tr>
<tr>
<td>795*</td>
<td>F</td>
<td>N/A</td>
</tr>
<tr>
<td>796*</td>
<td>exclamation point</td>
<td>N/A</td>
</tr>
<tr>
<td>797*</td>
<td>lower D</td>
<td>N/A</td>
</tr>
<tr>
<td>798*</td>
<td>lower E</td>
<td>N/A</td>
</tr>
<tr>
<td>799*</td>
<td>lower F</td>
<td>N/A</td>
</tr>
<tr>
<td>800*</td>
<td>g</td>
<td>N/A</td>
</tr>
<tr>
<td>801*</td>
<td>h</td>
<td>N/A</td>
</tr>
<tr>
<td>802*</td>
<td>I</td>
<td>N/A</td>
</tr>
<tr>
<td>803*</td>
<td>star</td>
<td>N/A</td>
</tr>
<tr>
<td>804*</td>
<td>lower G</td>
<td>N/A</td>
</tr>
<tr>
<td>805*</td>
<td>lower H</td>
<td>N/A</td>
</tr>
<tr>
<td>806*</td>
<td>lower I</td>
<td>N/A</td>
</tr>
<tr>
<td>807*</td>
<td>j</td>
<td>N/A</td>
</tr>
<tr>
<td>808*</td>
<td>k</td>
<td>N/A</td>
</tr>
<tr>
<td>809*</td>
<td>l</td>
<td>N/A</td>
</tr>
<tr>
<td>810*</td>
<td>pound (or hash)</td>
<td>N/A</td>
</tr>
<tr>
<td>811*</td>
<td>lower J</td>
<td>N/A</td>
</tr>
<tr>
<td>812*</td>
<td>lower K</td>
<td>N/A</td>
</tr>
<tr>
<td>813*</td>
<td>lower L</td>
<td>N/A</td>
</tr>
<tr>
<td>814*</td>
<td>m</td>
<td>N/A</td>
</tr>
<tr>
<td>815*</td>
<td>n</td>
<td>N/A</td>
</tr>
<tr>
<td>816*</td>
<td>o</td>
<td>N/A</td>
</tr>
<tr>
<td>818*</td>
<td>lower M</td>
<td>N/A</td>
</tr>
<tr>
<td>819*</td>
<td>lower N</td>
<td>N/A</td>
</tr>
<tr>
<td>820*</td>
<td>lower O</td>
<td>N/A</td>
</tr>
<tr>
<td>821*</td>
<td>p</td>
<td>N/A</td>
</tr>
<tr>
<td>822*</td>
<td>q</td>
<td>N/A</td>
</tr>
<tr>
<td>823*</td>
<td>r</td>
<td>N/A</td>
</tr>
<tr>
<td>824*</td>
<td>s</td>
<td>N/A</td>
</tr>
<tr>
<td>825*</td>
<td>lower P</td>
<td>N/A</td>
</tr>
</tbody>
</table>
### Table 54: Voice Prompts: Non-ASR and ASR (Continued)

<table>
<thead>
<tr>
<th>Prompt ID</th>
<th>Content (Non-ASR)</th>
<th>Content (ASR)</th>
</tr>
</thead>
<tbody>
<tr>
<td>826*</td>
<td>lower Q</td>
<td>N/A</td>
</tr>
<tr>
<td>827*</td>
<td>lower R</td>
<td>N/A</td>
</tr>
<tr>
<td>828*</td>
<td>lower S</td>
<td>N/A</td>
</tr>
<tr>
<td>829*</td>
<td>T</td>
<td>N/A</td>
</tr>
<tr>
<td>830*</td>
<td>u</td>
<td>N/A</td>
</tr>
<tr>
<td>831*</td>
<td>v</td>
<td>N/A</td>
</tr>
<tr>
<td>832*</td>
<td>question mark</td>
<td>N/A</td>
</tr>
<tr>
<td>833*</td>
<td>lower T</td>
<td>N/A</td>
</tr>
<tr>
<td>834*</td>
<td>lower U</td>
<td>N/A</td>
</tr>
<tr>
<td>835*</td>
<td>lower V</td>
<td>N/A</td>
</tr>
<tr>
<td>836*</td>
<td>w</td>
<td>N/A</td>
</tr>
<tr>
<td>837*</td>
<td>x</td>
<td>N/A</td>
</tr>
<tr>
<td>838*</td>
<td>Y</td>
<td>N/A</td>
</tr>
<tr>
<td>839*</td>
<td>z</td>
<td>N/A</td>
</tr>
<tr>
<td>840*</td>
<td>lower W</td>
<td>N/A</td>
</tr>
<tr>
<td>841*</td>
<td>lower X</td>
<td>N/A</td>
</tr>
<tr>
<td>842*</td>
<td>lower Y</td>
<td>N/A</td>
</tr>
<tr>
<td>843*</td>
<td>lower Z</td>
<td>N/A</td>
</tr>
<tr>
<td>844*</td>
<td>colon</td>
<td>N/A</td>
</tr>
<tr>
<td>845*</td>
<td>period</td>
<td>N/A</td>
</tr>
<tr>
<td>846*</td>
<td>comma</td>
<td>N/A</td>
</tr>
<tr>
<td>847*</td>
<td>forward slash</td>
<td>N/A</td>
</tr>
<tr>
<td>848*</td>
<td>backward slash</td>
<td>N/A</td>
</tr>
<tr>
<td>849*</td>
<td>plus sign</td>
<td>N/A</td>
</tr>
<tr>
<td>850</td>
<td>Please hold while I check your e-mail.</td>
<td>N/A</td>
</tr>
<tr>
<td>851</td>
<td>...one unreadable message</td>
<td>N/A</td>
</tr>
<tr>
<td>852</td>
<td>...unreadable messages</td>
<td>N/A</td>
</tr>
<tr>
<td>853</td>
<td>You have no unreadable messages</td>
<td>N/A</td>
</tr>
<tr>
<td>854</td>
<td>To manage unreadable messages, press six</td>
<td>N/A</td>
</tr>
<tr>
<td>855</td>
<td>unreadable message...</td>
<td>N/A</td>
</tr>
<tr>
<td>856</td>
<td>If you are finished, please press pound (or hash). Otherwise, press the next key to continue spelling.</td>
<td>N/A</td>
</tr>
<tr>
<td>857</td>
<td>Your quota grace limit has been reached.</td>
<td>N/A</td>
</tr>
<tr>
<td>858</td>
<td>Please hold while I check your messages.</td>
<td>N/A</td>
</tr>
<tr>
<td>859</td>
<td>The message could not be played.</td>
<td>N/A</td>
</tr>
<tr>
<td>Prompt ID</td>
<td>Content (Non-ASR)</td>
<td>Content (ASR)</td>
</tr>
<tr>
<td>-----------</td>
<td>----------------------------------------------------------------------------------</td>
<td>------------------------------------------------------------------------------</td>
</tr>
<tr>
<td>860</td>
<td>Not used</td>
<td>Not used</td>
</tr>
<tr>
<td>861</td>
<td>I’m sorry there are no network mailboxes available for Node...</td>
<td>N/A</td>
</tr>
<tr>
<td>862</td>
<td>I’m sorry. This message cannot be forwarded.</td>
<td>N/A</td>
</tr>
<tr>
<td>863</td>
<td>I could not connect to any of your remote servers. You will be redirected to the local voice mail view.</td>
<td>N/A</td>
</tr>
<tr>
<td>864 - 875</td>
<td>Not used</td>
<td>Not used</td>
</tr>
<tr>
<td>876</td>
<td>To record a standard greeting, press one.</td>
<td>To record a standard greeting, say “Standard” or press one.</td>
</tr>
<tr>
<td>877</td>
<td>To copy the standard greeting to the speech-recognition greeting, press two.</td>
<td>To copy the standard greeting to the speech-recognition greeting, say “Copy” or press two.</td>
</tr>
<tr>
<td>878</td>
<td>To record a speech-recognition greeting, press three.</td>
<td>To record a speech-recognition greeting say “Speech Recognition” or press three.</td>
</tr>
<tr>
<td>879</td>
<td>If you would like to copy this recording to the speech-recognition greeting, press one.</td>
<td>If you would like to use this recording as the speech-recognition greeting, say “Copy” or press one.</td>
</tr>
<tr>
<td>880</td>
<td>To record a new speech-recognition greeting, press three.</td>
<td>To record a new speech-recognition greeting, say “New” or press three.</td>
</tr>
<tr>
<td>881</td>
<td>ñ (pronunciation: en-yay)</td>
<td>N/A</td>
</tr>
<tr>
<td>882</td>
<td>I’m sorry. This system prompt is not available for replacement.</td>
<td>N/A</td>
</tr>
<tr>
<td>883</td>
<td>The following message could not be delivered to...</td>
<td>N/A</td>
</tr>
<tr>
<td>884</td>
<td>To disable Speech Recognition mailbox commands for this call, press eight.</td>
<td>To disable Speech Recognition mailbox commands for this call, say ‘Disable’ or press eight.</td>
</tr>
<tr>
<td>885</td>
<td>Speech Recognition disabled.</td>
<td>N/A</td>
</tr>
<tr>
<td>886</td>
<td>I could not create a secure connection to your remote server. You will be redirected to the local voice mail view.</td>
<td>N/A</td>
</tr>
<tr>
<td>887</td>
<td>Please contact your system administrator.</td>
<td>N/A</td>
</tr>
<tr>
<td>888</td>
<td>E-Mail Reader was unable to retrieve your faxes from your e-mail server due to a login failure.</td>
<td>N/A</td>
</tr>
<tr>
<td>889</td>
<td>E-Mail Reader was unable to retrieve your e-mails from your e-mail server due to a login failure.</td>
<td>N/A</td>
</tr>
<tr>
<td>890</td>
<td>Please enter your e-mail account password. This is case sensitive. Using the keypad, you can scroll through symbols, numbers, and letters, both lower and upper case.</td>
<td>Please spell or enter your e-mail account password. This is case sensitive. You can speak symbols, numbers, and letters. For upper case letters, say “Upper” followed by the letter.</td>
</tr>
<tr>
<td>891</td>
<td>Using the keys on your telephone, please spell the last name of the person. Then press pound (or hash). For the letter Q, press seven. For the letter Z, press nine.</td>
<td>Please say the last name of the person you are trying to reach. Otherwise, using the keys on your telephone, please spell the last name of the person, then press pound (or hash). For the letter Q, press seven. For the letter Z, press nine.</td>
</tr>
<tr>
<td>Prompt ID</td>
<td>Content (Non-ASR)</td>
<td>Content (ASR)</td>
</tr>
<tr>
<td>----------</td>
<td>-------------------------------------------------</td>
<td>------------------------------------------------------------------------------</td>
</tr>
<tr>
<td>892</td>
<td>You have no voice mails</td>
<td>N/A</td>
</tr>
<tr>
<td>893</td>
<td>You have no e-mails</td>
<td>N/A</td>
</tr>
<tr>
<td>894</td>
<td>You have no faxes</td>
<td>N/A</td>
</tr>
<tr>
<td>895</td>
<td>...Network extension...</td>
<td>N/A</td>
</tr>
<tr>
<td>896</td>
<td>To enable Speech Recognition mailbox commands for this call, press eight.</td>
<td>To enable Speech Recognition mailbox commands for this call, say “Enable” or press eight.</td>
</tr>
<tr>
<td>897</td>
<td>There was an error accessing your messages; you will be returned to the main menu</td>
<td>N/A</td>
</tr>
<tr>
<td>898</td>
<td>...Network Mailbox...</td>
<td>N/A</td>
</tr>
<tr>
<td>899</td>
<td>N/A</td>
<td>I think you said “delete.” If this is correct, please press one; if this is not correct, please press two.</td>
</tr>
<tr>
<td>900</td>
<td>Speech Recognition enabled.</td>
<td>N/A</td>
</tr>
<tr>
<td>901</td>
<td>No days selected.</td>
<td>No days selected.</td>
</tr>
<tr>
<td>902</td>
<td>On</td>
<td>On</td>
</tr>
<tr>
<td>903</td>
<td>Off</td>
<td>Off</td>
</tr>
<tr>
<td>904 - 906</td>
<td>Not used</td>
<td>Not used</td>
</tr>
<tr>
<td>907</td>
<td>Message dated…</td>
<td>Message dated…</td>
</tr>
<tr>
<td>908</td>
<td>E-Mail Reader was unable to retrieve your faxes and e-mails from your e-mail server due to a login failure.</td>
<td>E-Mail Reader was unable to retrieve your faxes and e-mails from your e-mail server due to a login failure.</td>
</tr>
<tr>
<td>909</td>
<td>To copy this message to additional mailboxes, press four.</td>
<td>To copy this message to additional mailboxes, say “Copy” or press four.</td>
</tr>
<tr>
<td>910</td>
<td>To spell a new name, press four.</td>
<td>To spell a new name, say “new name,” or press four.</td>
</tr>
<tr>
<td>911</td>
<td>Using the keys on your telephone, please spell the first name of the person. Then press pound (or hash). For the letter Q, press seven. For the letter Z, press nine.</td>
<td>Please say the full name of the person you are trying to reach. Otherwise, using the keys on your telephone, please spell the first name of the person, then press pound (or hash). For the letter Q, press seven. For the letter Z, press nine.</td>
</tr>
<tr>
<td>912</td>
<td>Using the keys on your telephone, please spell the last name of the person. Then press pound (or hash). For the letter Q, press seven. For the letter Z, press nine.</td>
<td>Please say the full name of the person you are trying to reach? Otherwise, using the keys on your telephone, please spell the last name of the person, then press pound (or hash). For the letter Q, press seven. For the letter Z, press nine.</td>
</tr>
<tr>
<td>913</td>
<td>When you are finished entering mailbox numbers, press star to exit.</td>
<td>When you are finished entering mailbox numbers, say “Finished” or press star to exit.</td>
</tr>
<tr>
<td>914</td>
<td>Enter the destination mailbox or Group list number. For the Directory, press pound (or hash). When you are finished entering the destination, press star to exit</td>
<td>Say or enter the destination mailbox or group list number. For the directory, say “Directory” or press pound (or hash). When you are finished entering the destination press star to exit.</td>
</tr>
<tr>
<td>915</td>
<td>...or if you are finished listening to the message, Hang up.</td>
<td>N/A</td>
</tr>
</tbody>
</table>
### Table 54: Voice Prompts: Non-ASR and ASR (Continued)

<table>
<thead>
<tr>
<th>Prompt ID</th>
<th>Content (Non-ASR)</th>
<th>Content (ASR)</th>
</tr>
</thead>
<tbody>
<tr>
<td>916</td>
<td>Subscriber access is not programmed yet. Please contact the system administrator.</td>
<td>N/A</td>
</tr>
<tr>
<td>917</td>
<td>I am sorry, I could not complete your call at this time. Please try again later.</td>
<td>N/A</td>
</tr>
<tr>
<td>918</td>
<td>Message access is not configured. Please contact your system administrator.</td>
<td>N/A</td>
</tr>
<tr>
<td>919</td>
<td>To change your main menu e-mail reader count, press four.</td>
<td>To change your main menu e-mail reader count, say e-mail reader count or press four.</td>
</tr>
<tr>
<td>920</td>
<td>Main Menu E-Mail Reader Count is currently enabled.</td>
<td>N/A</td>
</tr>
<tr>
<td>921</td>
<td>Main Menu E-Mail Reader Count is currently disabled.</td>
<td>N/A</td>
</tr>
<tr>
<td>922</td>
<td>To enable, press one.</td>
<td>To enable, say “enable” or press one.</td>
</tr>
<tr>
<td>923</td>
<td>To disable, press one.</td>
<td>To disable, say “disable” or press one.</td>
</tr>
<tr>
<td>924</td>
<td>Main Menu E-Mail Reader Count has been automatically disabled, please go to the personal options menu to re-enable this feature.</td>
<td>N/A</td>
</tr>
<tr>
<td>926</td>
<td>For help with password special keys, press two.</td>
<td>For help with password special keys, say “Help” or press two.</td>
</tr>
<tr>
<td>927</td>
<td>To repeat the following instructions press zero at any time.</td>
<td>To repeat the following instructions say “Repeat” or press zero at any time.</td>
</tr>
<tr>
<td>928</td>
<td>Enter the destination mailbox or Group list number. When you are finished entering the destination, press star to exit.</td>
<td>Say or enter the destination mailbox or group list number. When you are finished entering the destination, press star to exit.</td>
</tr>
<tr>
<td>929</td>
<td>“equal to sign”</td>
<td>N/A</td>
</tr>
<tr>
<td>930</td>
<td>“greater than sign”</td>
<td>N/A</td>
</tr>
<tr>
<td>931</td>
<td>“less than sign”</td>
<td>N/A</td>
</tr>
<tr>
<td>932</td>
<td>“semi colon sign”</td>
<td>N/A</td>
</tr>
<tr>
<td>933</td>
<td>“percent sign”</td>
<td>N/A</td>
</tr>
<tr>
<td>934</td>
<td>“dollar sign”</td>
<td>N/A</td>
</tr>
<tr>
<td>935</td>
<td>“pipe sign”</td>
<td>N/A</td>
</tr>
<tr>
<td>936</td>
<td>“pound (or hash) sign”</td>
<td>N/A</td>
</tr>
<tr>
<td>937</td>
<td>I am sorry, you must enter a password for security reasons. Please, enter your new password, then press pound (or hash).</td>
<td>N/A</td>
</tr>
<tr>
<td>938</td>
<td>To listen to all additional messages, please press eight.</td>
<td>Say “All Messages” or press eight to listen to all additional messages.</td>
</tr>
</tbody>
</table>
### Table 54: Voice Prompts: Non-ASR and ASR (Continued)

<table>
<thead>
<tr>
<th>Prompt ID</th>
<th>Content (Non-ASR)</th>
<th>Content (ASR)</th>
</tr>
</thead>
<tbody>
<tr>
<td>939</td>
<td>The locations of special characters are... For space, at, hyphen, underscore, equals sign, comma, greater than, and right parenthesis, use key “zero.” For exclamation point, pipe, apostrophe, colon, semi-colon, forward slash, and backslash, use key “one.” For period, tilde, dollar, asterisk, ampersand, percentage, plus, less than and left parenthesis, use key “star.” For pound (or hash), use key “pound (or hash).”</td>
<td>N/A</td>
</tr>
<tr>
<td>940</td>
<td>This call routing application does not have an associated digit translation node. Please assign a digit translation node using the System Manager interface before attempting to assign an Audiotex greeting using the System Administrator interface.</td>
<td>N/A</td>
</tr>
<tr>
<td>941</td>
<td>E-Mail Reader encountered an error while trying to connect to your e-mail server. Please contact your system administrator for further assistance.</td>
<td>N/A</td>
</tr>
<tr>
<td>942</td>
<td>E-mail Reader was unable to connect to your e-mail server. The server may be down or there may be a configuration issue with your E-Mail Reader Profile. Please contact your system administrator for further assistance.</td>
<td>N/A</td>
</tr>
<tr>
<td>943</td>
<td>E-Mail Reader cannot connect to your e-mail server due to an encryption mismatch. Please contact your system administrator for further assistance.</td>
<td>N/A</td>
</tr>
<tr>
<td>944</td>
<td>I am sorry, your password must be at least four digits long. Please, enter your new password, then press pound.</td>
<td>N/A</td>
</tr>
</tbody>
</table>
Endpoint User Guides

Table 55 lists the Inter-Tel endpoint user guides that include instructions for using Inter-Tel 5000 features. Detailed procedures for particular endpoints appear in the current version of the affected user guide.

**Table 55. Inter-Tel Endpoint User Guides**

<table>
<thead>
<tr>
<th>User Guide</th>
<th>Part Number</th>
</tr>
</thead>
<tbody>
<tr>
<td>Model 8600 User Guide – Inter-Tel Protocol Mode</td>
<td>550.8115</td>
</tr>
<tr>
<td>Model 8620/8622 User Guide – Inter-Tel Protocol Mode</td>
<td>550.8119</td>
</tr>
<tr>
<td>Model 8662 User Guide – Inter-Tel Protocol Mode</td>
<td>550.8117</td>
</tr>
<tr>
<td>Model 8690 User Guide – Inter-Tel Protocol Mode</td>
<td>550.8116</td>
</tr>
<tr>
<td>Model 8560 (Digital) and Model 8660 (IP) User Guide</td>
<td>550.8113</td>
</tr>
<tr>
<td>Model 8520 (Digital) User Guide</td>
<td>550.8112</td>
</tr>
<tr>
<td>Model 8500 (Digital) User Guide</td>
<td>550.8114</td>
</tr>
<tr>
<td>Executive Display Endpoint (Digital)/IP PhonePlus Endpoint User Guide</td>
<td>550.8108</td>
</tr>
<tr>
<td>Professional Display Endpoint (Digital)/IP PhonePlus Endpoint User Guide</td>
<td>560.8002</td>
</tr>
<tr>
<td>Standard Display and Basic Digital Endpoint User Guide</td>
<td>550.8109</td>
</tr>
<tr>
<td>Associate Display and Basic Digital Endpoint User Guide</td>
<td>560.8003</td>
</tr>
<tr>
<td>Model 8664/8665/8668 User Guide</td>
<td>935.4522</td>
</tr>
<tr>
<td>Model 8524/8525/8526 Wireless Endpoint User Guide</td>
<td>935.4519</td>
</tr>
<tr>
<td>Model 8602 IP Softphone</td>
<td>835.3019</td>
</tr>
</tbody>
</table>
Abbreviations and Acronyms

In addition to industry-wide and/or internationally recognized units of measure, such as Hz, V, AC, DC, A, ms, etc., the abbreviations and acronyms shown in Table 56 appear in this document.

Table 56. Abbreviations and Acronyms

<table>
<thead>
<tr>
<th>Abbreviation</th>
<th>Spelled-Out Meaning</th>
</tr>
</thead>
<tbody>
<tr>
<td>ACD</td>
<td>Automatic Call Distribution</td>
</tr>
<tr>
<td>AMIS</td>
<td>Audio Messaging Interchange Specification</td>
</tr>
<tr>
<td>ANI</td>
<td>Automatic Number Identification</td>
</tr>
<tr>
<td>ANSI</td>
<td>American National Standards Institute</td>
</tr>
<tr>
<td>ARS</td>
<td>Automatic Route Selection</td>
</tr>
<tr>
<td>ASAI</td>
<td>Adjunct Switch Application Interface</td>
</tr>
<tr>
<td>ASR</td>
<td>Automatic Speech Recognition</td>
</tr>
<tr>
<td>AVDAP</td>
<td>Advanced Voice Data Applications Processor</td>
</tr>
<tr>
<td>BLF</td>
<td>Busy Lamp Field</td>
</tr>
<tr>
<td>BRM</td>
<td>Basic Rate Module</td>
</tr>
<tr>
<td>BVM</td>
<td>Basic Voice Mail</td>
</tr>
<tr>
<td>CF</td>
<td>Compact Flash</td>
</tr>
<tr>
<td>CLID</td>
<td>Calling Line Identification [European/British]</td>
</tr>
<tr>
<td>CO</td>
<td>Central Office</td>
</tr>
<tr>
<td>CP</td>
<td>Call Processing</td>
</tr>
<tr>
<td>CS</td>
<td>Communication Server</td>
</tr>
<tr>
<td>CT</td>
<td>Computer Telephony</td>
</tr>
<tr>
<td>DB</td>
<td>Database</td>
</tr>
<tr>
<td>DDI</td>
<td>Direct Dialing Inward [DID in Europe]</td>
</tr>
<tr>
<td>DEI</td>
<td>Digital Expansion Interface</td>
</tr>
<tr>
<td>DEM</td>
<td>Digital Expansion Module</td>
</tr>
<tr>
<td>DID</td>
<td>Direct Inward Dialing</td>
</tr>
<tr>
<td>DISA</td>
<td>Direct Inward System Access</td>
</tr>
<tr>
<td>DND</td>
<td>Do-Not-Disturb</td>
</tr>
<tr>
<td>DNIS</td>
<td>Dialed Number Identification Service</td>
</tr>
<tr>
<td>DSP</td>
<td>Digital Signal Processor</td>
</tr>
<tr>
<td>DSS</td>
<td>Direct Station Selection</td>
</tr>
<tr>
<td>E&amp;M</td>
<td>Ear and Mouth</td>
</tr>
<tr>
<td>EM</td>
<td>Enterprise® Messaging</td>
</tr>
<tr>
<td>FCC</td>
<td>Federal Communications Commission</td>
</tr>
<tr>
<td>FoIP</td>
<td>Fax over Internet Protocol</td>
</tr>
<tr>
<td>IC</td>
<td>Intercom</td>
</tr>
<tr>
<td>IP</td>
<td>Internet Protocol</td>
</tr>
<tr>
<td>Abbreviation</td>
<td>Spelled-Out Meaning</td>
</tr>
<tr>
<td>--------------</td>
<td>-------------------------------------</td>
</tr>
<tr>
<td>IPR</td>
<td>IP Resource</td>
</tr>
<tr>
<td>IPRA</td>
<td>IPR Application</td>
</tr>
<tr>
<td>ISDN</td>
<td>Integrated Services Digital Network</td>
</tr>
<tr>
<td>ITP</td>
<td>Inter-Tel Protocol</td>
</tr>
<tr>
<td>LAN</td>
<td>Local Area Network</td>
</tr>
<tr>
<td>LCD</td>
<td>Liquid Crystal Display</td>
</tr>
<tr>
<td>LED</td>
<td>Light Emitting Diode</td>
</tr>
<tr>
<td>LSM</td>
<td>Loop Start Module</td>
</tr>
<tr>
<td>MGCP</td>
<td>Media Gateway Control Protocol</td>
</tr>
<tr>
<td>MOH</td>
<td>Music On Hold</td>
</tr>
<tr>
<td>NAT</td>
<td>Network Address Translation</td>
</tr>
<tr>
<td>N/A</td>
<td>Not Available or Not Applicable, as appropriate</td>
</tr>
<tr>
<td>NTP</td>
<td>Network Time Protocol</td>
</tr>
<tr>
<td>OAI</td>
<td>Open Architecture Interface</td>
</tr>
<tr>
<td>OPX</td>
<td>Off Premises Extension</td>
</tr>
<tr>
<td>OSE</td>
<td>Open Standards Edition</td>
</tr>
<tr>
<td>P2P</td>
<td>Peer to Peer</td>
</tr>
<tr>
<td>PC</td>
<td>Personal Computer</td>
</tr>
<tr>
<td>PCM</td>
<td>Pulse Code Modulation</td>
</tr>
<tr>
<td>PDA</td>
<td>Personal Digital Assistant</td>
</tr>
<tr>
<td>PDF</td>
<td>Portable Document Format</td>
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<td>TFTP</td>
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<td>UDP</td>
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<td>UK</td>
<td>United Kingdom</td>
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<td>UL</td>
<td>Underwriters Laboratories, Inc.</td>
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<td>U.S.</td>
<td>United States of America</td>
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Voice Mail Flowchart

NOTE
At any menu level, you can say “Cancel” or press  to cancel or return to the previous menu or say “Accept” or press  to accept. Also, Automatic Speech Recognition commands are indicated in the quotation marks.

ACCESSING THE SYSTEM ADMINISTRATOR’S MAILBOX

• EITHER, Press the MSG button to respond to a message from voice mail.
• OR, Call the voice mail extension number, say “Login” or press , and say or enter your mailbox number:
  • Enter your password (press ).
  • Say “Nine” or press  to access administrator features.

PERSONAL OPTIONS
Record “Greeting”  
“Primary” Greeting  1  
“Alternate” Greeting  2  
“System” Greeting  3  
(See Recording Options)
Record Directory “Name”  2  
(See Recording Options)
Change “Password”  3  
Change “Envelope” Settings *  4  
“Voice Mail”  1  
“E-Mail”  2  
“Facsimile”  3  
Then select: 
“Time, Date”  1  
“Source”  2  
“Length/Subj./Pages”  3  
“All Options”  4  
“Nope”  5  
“Remote” Messaging *  3  
“Primary” Cascade  1  
“Alternate” Cascade  2  
“More” Options  9  
Fax “Number”  1  
Message “Order”  2  
“E-Mail Rdr. Count”  4  
“Transfer” Method *  5  

TRANSFER METHOD *
“Unannounced”  1  
“Screened”  2  
“Announce” Only  3  
“Cancel”  4  

CREATE CUSTOM RECORDING
If existing recording plays:
“Standard” *  1  
“Copy” *  2  
“Speech Recognition” *  3  
Then select: 
“Replay”  1  
“Append”  2  
“Erase” and Re-record  3  
“Accept”  4  
If no recording exists, see “Making a Recording”

SENDING A MESSAGE
During your message:
Pause  2  
Erase and Re-record  3  
“Continue”  4  
After your message:
“Replay”  1  
“Append”  2  
“Erase”  3  
“Accept”  4  
Hang up to cancel

RECORDING OPTIONS
“Replay”  1  
“Append”  2  
“Erase” and Re-record  3  
“Accept”  4  

MESSAGE ORDER
“New” Messages  1  
“Saved” Messages  2  
Then set search order: 
“First In”  1  
“Last In”  3  

RECORD CUSTOM PROMPTS
“Standard” *  1  
“Speech Recognition” *  2  
Then select: 
“Replay”  1  
“Custom”  2  
“System”  3  
“Cancel”  4  
“Options”  5  

RECORDING OPTIONS
“Replay”  1  
“Append”  2  
“Erase” and Re-record  3  
“Cancel”  4  
“Accept”  5  

TRANSFER METHOD *
“Unannounced”  1  
“Screened”  2  
“Announce” Only  3  
“Cancel”  4  

IMPORT FAX DOCUMENT
Record “Greeting”  1  
“Primary” Greeting  2  
“Alternate” Greeting  3  
“System” Greeting  4  
(See Recording Options)
Record Directory “Name”  5  
(See Recording Options)
Change “Password”  6  
Change “Envelope” Settings *  7  
“Voice Mail”  1  
“E-Mail”  2  
“Facsimile”  3  
Then select: 
“Time, Date”  1  
“Source”  2  
“Length/Subj./Pages”  3  
“All Options”  4  
“Nope”  5  
“Remote” Messaging *  6  
“Primary” Cascade  1  
“Alternate” Cascade  2  
“More” Options  7  
Fax “Number”  1  
Message “Order”  2  
“E-Mail Rdr. Count”  3  
“Transfer” Method *  4  

REMOTE MESSAGING *
Program Cascade “Level”  1  
Enter cascade level number, then:
“Change”  1  
Enter “Personal” #  2  
Enter “Pager” #  3  
Enter “Time” of Day  2  
Set “Days” of Week  3  
Mon-Fri “Weekdays”  1  
All “Days”  2  
Individual “Day”  3  
Select Message “Category”  4  
“All Messages”  1  
“Priority” Messages  2  

* Menu options vary, based on available and programmable features. See the “Administrator Procedures” chapter for details.