



Cisco Unified IP Phone 7962G and 7942G Phone Guide for Cisco Unified Communications Manager 7.0 (SCCP and SIP)

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Common Phone Tasks

View online help on the phone	Press .
Place a call	Go off-hook before or after dialing a number.
Redial a number	Press Redial. Or press the Navigation button while on-hook to see your Placed Calls log.
Switch to the handset during a call	Pick up the handset.
Switch to the speaker or headset during a call	Press or o, then hang up the handset.
Mute your phone	Press .
Use your call logs	Press to choose a call log. To dial, highlight a listing and go off-hook.
Edit a number	Press EditDial, << or >>.
Hold/resume a call	Press Hold or Resume.
Transfer a call to a new number	Press Transfer, enter the number, then press Transfer again.
Place an intercom call	Press intercom button, enter a number if necessary, and speak after you hear the tone.
Start a standard (ad hoc) conference call	Press more > Confrn, dial the participant, then press Confrn again.



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QUICK REFERENCE



Cisco Unified IP Phone 7962G and 7942G for Cisco Unified Communications Manager 7.0 (SCCP and SIP)

Softkey Definitions Phone Screen Icons Button Icons Common Phone Tasks

Softkey Definitions

AbbrDial	Dial using a speed dial index number
Answer	Answer a call
Back	Return to the previous Help topic
Barge	Add yourself to a call on a shared line
CallBack	Receive notification when a busy extension becomes available
Cancel	Cancel an action or exit a screen without applying changes
cBarge	Add yourself to a call on a shared line and establish a conference
CFwdALL	Set up/cancel call forwarding
Clear	Delete records or settings
Close	Close the current window
ConfList	View conference participants
Confrn	Create a conference call
Delete	Remove characters to the right of the cursor when using EditDial
Details (SCCP only)	Open the Details record for a multiparty call in the Missed Calls and Received Calls logs
Dial	Dial a phone number
DirTrfr (SCCP only)	Transfer two calls to each other
DND	Turn on/off Do Not Disturb (DND)
EditDial	Edit a number in a call log
EndCall	Disconnect the current call or the current intercom call
Erase	Reset settings to their defaults
Exit	Return to the previous screen

GPickUp	Answer a call that is ringing in another group or on another line
iDivert	Send or redirect a call to a voice messaging system
Join	Join together existing calls to create a conference
Links	View related Help topics
Main	Display the Help main menu
MeetMe	Host a Meet-Me conference call
more	Display additional softkeys
New Call	Make a new call
OPickUp	Answer a call that is ringing in an associated group
Park	Store a call using Call Park
PickUp	Answer a call that is ringing on another phone in your group
QRT	Submit call problems to the system administrator
Redial	Redial the most recently dialed number
Remove	Remove a conference participant
Resume	Resume a call on hold
RmLstC	Drop the last party added to a conference call
Save	Save the chosen settings
Search	Search for a directory listing
Select	Select a menu item or call
Transfer	Transfer a call
Update	Refresh content
VidMode (SCCP only)	Choose a video display mode
<<	Delete entered characters
>>	Move through entered characters

Phone Screen Icons

	Call Forwarding enabled
6	Call on hold
€	Connected call
	Incoming call
E	Off-hook
=	On-hook
	Shared line in use
<i>©</i> ⊠	Message waiting
G	Authenticated call
€	Encrypted call
A	BLF-monitored line is in-use
⇔	BLF-monitored line is idle
	BLF-monitored line is ringing (BLF Pickup)
#	Speed-dial, call log, or directory listing (line status unknown)
Ö	Line in Do Not Disturb (BLF feature)
	Intercom line in idle state
3	Intercom line in one-way audio call
-	Intercom line in two-way audio call
G.	Handset in use
<u>ទី</u> <u>៤</u> ព	Headset in use
ц	Speakerphone in use

≕ i	Video enabled (SCCP only)
\triangleright	Feature assigned to button
	Mobility assigned to button
	Hold assigned to button
607	Conference assigned to button
R	Transfer assigned to button
3	Phone service URL assigned to button
@	URL entry in a call log is ready to edit (SIP only)
~	Option selected
9	Feature enabled

Button Icons

	Messages
(3)	Services
?	Help
THE STATE OF THE S	Directories
8	Settings
	Volume
	Speaker
2	Mute
0	Headset

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Getting Started

Using this Guide

This guide provides you with an overview of the features available on your phone. You can read it completely for a solid understanding of your phone's capabilities, or refer to the table below for pointers to commonly used sections.

If you want to	Then
Explore your phone on your own	Press on the phone when you need assistance.
Review safety information	See Safety and Performance Information, page 2.
Connect your phone	See Connecting Your Phone, page 4.
Use your phone after it is installed	Start with An Overview of Your Phone, page 7.
Learn what the button lights mean	See Understanding Buttons and Hardware, page 7.
Learn about the display	See Understanding Phone Screen Features, page 12.
Make calls	See Placing a Call—Basic Options, page 16.
Put calls on hold	See Using Hold and Resume, page 21.
Mute calls	See Using Mute, page 22.
Transfer calls	See Transferring Calls, page 24.
Make conference calls	See Making Conference Calls, page 28.
Set up speed dialing	See Speed Dialing, page 34.
Share a phone number	See Using a Shared Line, page 38.
Use your phone as a speakerphone	See Using a Handset, Headset, and Speakerphone, page 48.
Change the ring volume or tone	See Changing Phone Settings, page 51.
View your missed calls	See Using Call Logs and Directories, page 53.
Listen to your voice messages	See Accessing Voice Messages, page 59.
See softkey and icon definitions	See the Quick Reference Card in the front of this guide.

Finding Additional Information

You can access the most current Cisco Unified IP Phone documentation on the World Wide Web at this URL:

http://www.cisco.com/en/US/products/hw/phones/ps379/tsd_products_support_series_home.html

You can access the Cisco website at this URL:

http://www.cisco.com/

International Cisco websites can be accessed from this URL:

http://www.cisco.com/public/countries_languages.shtml

You can access the most current licensing information at this URL:

http://www.cisco.com/en/US/docs/voice_ip_comm/cuipph/all_models/openssl_license/7900_ssllic.htm

Accessing Cisco Unified IP Phone 7900 Series eLearning Tutorials

(SCCP phones only.)

The Cisco Unified IP Phone 7900 Series eLearning tutorials use audio and animation to demonstrate basic calling features. You can access eLearning tutorials online (for several phone models) from your personal computer. Look for the eLearning tutorial (English only) for your phone model in the documentation list at the following location:

http://www.cisco.com/en/US/products/hw/phones/ps379/products_user_guide_list.html



Although a eLearning tutorials for the Cisco Unified IP Phone 7962G and 7942G are not available, refer to the Cisco Unified IP Phone 7900 Series eLearning tutorials for an overview of the common IP phone features and functionality.

Safety and Performance Information

Refer to these sections for information about the impact of power outages and other devices on your Cisco Unified IP Phone.

Power Outage

Your accessibility to emergency service through the phone is dependent on the phone being powered. If there is an interruption in the power supply, Service and Emergency Calling Service dialing will not function until power is restored. In the case of a power failure or disruption, you may need to reset or reconfigure equipment before using the Service or Emergency Calling Service dialing.

Using External Devices

The following information applies when you use external devices with the Cisco Unified IP Phone:

Cisco recommends the use of good quality external devices (such as headsets) that are shielded against unwanted radio frequency (RF) and audio frequency (AF) signals.

Depending on the quality of these devices and their proximity to other devices such as mobile phones or two-way radios, some audio noise may still occur. In these cases, Cisco recommends that you take one or more of the following actions:

- Move the external device away from the source of the RF or AF signals.
- Route the external device cables away from the source of the RF or AF signals.
- Use shielded cables for the external device, or use cables with a better shield and connector.
- Shorten the length of the external device cable.
- Apply ferrites or other such devices on the cables for the external device.

Cisco cannot guarantee the performance of the system because Cisco has no control over the quality of external devices, cables, and connectors. The system will perform adequately when suitable devices are attached using good quality cables and connectors.



In European Union countries, use only external headsets that are fully compliant with the EMC Directive [89/336/EC].

Cisco Product Security Overview

This product contains cryptographic features and is subject to United States and local country laws governing import, export, transfer and use. Delivery of Cisco cryptographic products does not imply third-party authority to import, export, distribute or use encryption. Importers, exporters, distributors and users are responsible for compliance with U.S. and local country laws. By using this product you agree to comply with applicable laws and regulations. If you are unable to comply with U.S. and local laws, return this product immediately.

Further information regarding U.S. export regulations may be found at http://www.access.gpo.gov/bis/ear/ear_data.html.

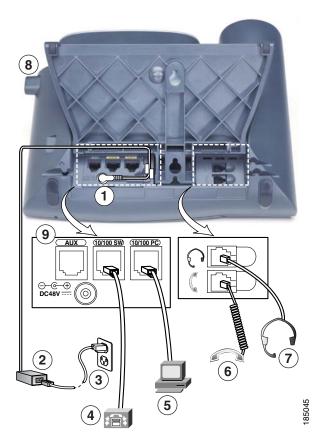
Accessibility Features

A list of accessibility features is available upon request. You can find more information about Cisco accessibility features at this URL:

www.cisco.com/go/accessibility

Connecting Your Phone

Your system administrator will likely connect your new Cisco Unified IP Phone to the corporate IP telephony network. If that is not the case, refer to the graphic and table below to connect your phone.



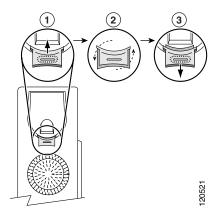
1	DC adaptor port (DC48V)	6	Handset port
2	AC-to-DC power supply	7	Headset port
3	AC power cord	8	Footstand button
4	Network port (10/100 SW)	9	Auxiliary port (AUX)
5	Access port (10/100 PC)		

Adjusting the Footstand

To change the angle of the phone base, adjust the footstand while pressing the footstand button.

Adjusting the Handset Rest

Cisco recommends adjusting the handset rest, particularly when wall mounting the phone, as this will ensure that the receiver will not readily slip out of the cradle. See the table below for instructions.



- 1 Set the handset aside and pull the square plastic tab from the handset rest.
- **2** Rotate the tab 180 degrees.
- 3 Slide the tab back into the handset rest. An extension protrudes from the top of the rotated tab. Return the handset to the handset rest.

Registering with TAPS

After your phone is connected to the network, your system administrator might ask you to auto-register your phone using TAPS (Tool for Auto-Registered Phones Support). TAPS might be used for a new phone or to replace an existing phone.

To register with TAPS, pick up the handset, enter the TAPS extension provided by your system administrator, and follow the voice prompts. You might need to enter your entire extension, including the area code. After your phone displays a confirmation message, hang up. The phone will re-start.

Using a Headset

You can use a wired headset, or you can use a wireless headset in conjunction with the wireless headset remote hookswitch control feature.

Headset Support

Although Cisco Systems performs limited internal testing of third-party headsets for use with the Cisco Unified IP Phones, Cisco does not certify or support products from headset (or handset) vendors.

Cisco recommends the use of good quality external devices, for example, headsets that are screened against unwanted radio frequency (RF) and audio frequency (AF) signals. Depending on the quality of these devices and their proximity to other devices such as mobile phones and two-way radios, some audio noise or echo may still occur. An audible hum or buzz may be heard by either the remote party or by both the remote party and the Cisco Unified IP Phone user. Humming or buzzing sounds can be caused by a range of outside sources: for example, electric lights, electric motors, or large PC monitors. See Using External Devices, page 3 for more information.



In some cases, hum may be reduced or eliminated by using a local power cube or power injector.

These environmental and hardware inconsistencies in the locations where Cisco Unified IP Phones are deployed means that there is not a single headset solution that is optimal for all environments.

Cisco recommends that customers test headsets in their intended environment to determine performance before making a purchasing decision and deploying en masse.

Audio Quality Subjective to the User

Beyond the physical, mechanical and technical performance, the audio portion of a headset must sound good to the user and to the party on the far end. Sound quality is subjective and Cisco cannot guarantee the performance of any headsets. However, a variety of headsets from leading headset manufacturers have been reported to perform well with Cisco Unified IP Phones. See manufacturer's sites for details.

For information about wireless headsets that work in conjunction with the wireless headset remote hookswitch control feature, got to the following URL: http://www.cisco.com/pcgi-bin/ctdp/Search.pl

- **1.** Choose **IP Communications** from the Enter Solution drop-down list box. The Select a Solution Category drop-down list box displays.
- 2. Choose IP Phone Headsets to see a list of Technology Development Program partners.

If you want to search for a particular Technology Development Program partner, enter the partner's name in the Enter Company Name box.

An Overview of Your Phone

The Cisco Unified IP Phone 7962G and 7942G are full-feature telephones that provide voice communication over the same data network that your personal computer uses, allowing you to place and receive phone calls, put calls on hold, speed dial numbers, transfer calls, make conference calls, and so on.

In addition to basic call-handling features, your phone can provide enhanced productivity features that extend your call-handling capabilities. Depending on configuration, your phone supports:

- Access to network data, XML applications, and web-based services.
- Online customizing of phone features and services from your Cisco Unified CM User Options web pages.
- A comprehensive online help system that displays information on the phone screen.

Understanding Buttons and Hardware

You can use the graphics and table below to identify buttons and hardware on your phone.

Cisco Unified IP Phone 7962G

Cisco Unified IP Phone 7942G

	Item	Description	For more information, see
1	Programmable buttons	Depending on configuration, programmable buttons provide access to: • Phone lines and intercom lines (line buttons) and intercom lines • Speed-dial numbers (speed-dial buttons, including the BLF speed-dial feature) • Web-based services (for example, a Personal Address Book button) • Call features (for example, a Privacy, Hold, or Transfer button) Buttons illuminate to indicate status: Green, steady—Active call or two-way intercom call Green, flashing—Held call Amber, steady—Privacy in use, one-way intercom call, DND active, or logged into Hunt Group Amber, flashing—Incoming call or reverting call Red, steady—Remote line in use (shared line or BLF status)	 Understanding Phone Screen Features, page 12 Basic Call Handling, page 16 Speed Dialing, page 34 Using a Shared Line, page 38 Using BLF to Determine a Line State, page 41 Using BLF to Determine a Line State, page 41 Placing or Receiving Intercom Calls, page 32
2	Phone screen	Shows phone features.	Understanding Phone Screen Features, page 12
3	Footstand button	Allows you to adjust the angle of the phone base.	Adjusting the Footstand, page 5
4	Messages button	Auto-dials your voice message service (varies by service).	Accessing Voice Messages, page 59
5	Directories button	Opens/closes the Directories menu. Use it to access call logs and directories.	Using Call Logs, page 53
6	Help button	Activates the Help menu.	Accessing the Help System on Your Phone, page 13

	Item	Description	For more information, see
7	Settings button	Opens/closes the Settings menu. Use it to control phone screen contrast and ring sounds.	Changing Phone Settings, page 51
8	Services button	Opens/closes the Services menu.	Using the User Options Web Pages, page 60
9	Volume button	Controls the handset, headset, and speakerphone volume (off-hook) and the ringer volume (on-hook).	Using a Handset, Headset, and Speakerphone, page 48
10	Speaker button	Toggles the speakerphone on or off. When the speakerphone is on, the button is lit.	Using a Handset, Headset, and Speakerphone, page 48
11	Mute button	Toggles the microphone on or off. When the microphone is muted, the button is lit.	Using Mute, page 22
12	Headset button	Toggles the headset on or off. When the headset is on, the button is lit.	Using a Handset, Headset, and Speakerphone, page 48
13	Navigation button	Allows you to scroll through menus and highlight items. When the phone is on-hook, displays phone numbers from your Placed Calls log.	Using Call Logs, page 53
14	Keypad	Allows you to dial phone numbers, enter letters, and choose menu items.	Basic Call Handling, page 16
15	Softkey buttons	Each activates a softkey option (displayed on your phone screen).	Understanding Phone Screen Features, page 12
16	Handset light strip	Indicates an incoming call or new voice message.	Accessing Voice Messages, page 59

Understanding Lines and Calls

To avoid confusion about lines and calls, refer to these descriptions:

Lines—Each line corresponds to a directory number or intercom number that others can use to call you. The Cisco Unified IP Phone 7962G supports one to six lines and the Cisco Unified IP Phone 7942G supports one to two lines, depending on configuration. To see how many lines you have, look at the right side of your phone screen. You have as many lines as you have directory numbers and phone line icons:

Calls—Each line can support multiple calls. By default, your phone supports four connected calls per line, but your system administrator can adjust this number according to your needs. Only one call can be active at any time; other calls are automatically placed on hold.

Understanding Line and Call Icons

Your phone displays icons to help you determine the call and line state.

Icon	Line or Call State	Description
=	On-hook line	No call activity on this line.
E	Off-hook line	You are dialing a number or an outgoing call is ringing.
6	Connected call	You are currently connected to the other party.
2	Ringing call	An incoming call is ringing on one of your lines.
6	Call on hold	You have put the call on hold. See Using Hold and Resume, page 21.
€	Remote-in-use	Another phone that shares your line has a connected call. See Using a Shared Line, page 38, for details.
	Reverting call	A holding call is reverting to your phone. See Using Hold and Resume, page 21.
G ₀	Authenticated call	See Making and Receiving Secure Calls, page 42.
€	Encrypted call	See Making and Receiving Secure Calls, page 42.
⇔	BLF- monitored line is idle	See Using BLF to Determine a Line State, page 41.
E	BLF- monitored line is in-use	See Using BLF to Determine a Line State, page 41.
(AZA	BLF-monitored line is ringing (BLF Pickup)	See Using BLF to Determine a Line State, page 41.
Ö	Line in Do Not Disturb (BLF)	See Using BLF to Determine a Line State, page 41.
≣	Idle Intercom line	The intercom line is not in use. See Placing or Receiving Intercom Calls, page 32.
5	One-way intercom call	The intercom line is sending or receiving one-way audio. See Placing or Receiving Intercom Calls, page 32.
₿	Two-way intercom call	The recipient pressed the intercom line to activate two-way audio with the caller. See Placing or Receiving Intercom Calls, page 32.

Understanding Phone Screen Features

This is what your main phone screen might look like with active calls and several feature menus open:

1	Primary phone line	Displays the phone number (directory number) for your primary phone line. When several feature tabs are open, the phone number and the time and date alternate display in this area.
2	Programmable button indicators	Programmable buttons and can serve as phone line buttons, speed-dial buttons, phone service buttons or phone feature buttons. Icons and labels indicate how these buttons are configured. For an icon reference, see "Phone Screen Icons" in the Quick Reference Card at the front of this guide.
3	Softkey labels	Each displays a softkey function. To activate a softkey, press the softkey button
4	Status line	Displays audio mode icons, status information, and prompts.
5	Call activity area	Displays current calls per line, including caller ID, call duration, and call state for the highlighted line (standard view). See "Understanding Line and Call Icons" section on page 11, and Viewing Multiple Calls, page 23.
6	Phone tab	Indicates call activity.
7	Feature tabs	Each indicates an open feature menu. See Understanding Feature Buttons and Menus, page 13.

Cleaning the Phone Screen

Use only a soft, dry cloth to wipe the phone screen. Do not use any liquids or powders on the phone, as they can contaminate phone components and cause failures.

Understanding Feature Buttons and Menus

Press a feature button to open or close a feature menu.

If you want to	Then	
Open or close a feature menu	Press a feature button: Messages Services Directories Settings Help	
Scroll through a list or menu	Press the Navigation button.	
Go back one level in a feature menu	Press Exit. Pressing Exit from the top level of a menu, closes the menu.	
Switch among open feature menus	Press a feature tab. Each feature menu has a corresponding tab. The tab is visible when the feature menu is open.	

Accessing the Help System on Your Phone

Your phone provides a comprehensive online help system. Help topics appear on the phone screen.

If you want to	Then
View the main menu	Press on your phone and wait a few seconds for the menu to display.
	Main menu topics include:
	About Your Cisco Unified IP Phone—Details about your phone
	How do I?—Procedures for common phone tasks
	Calling Features—Descriptions and procedures for calling features
	Help—Tips on using and accessing Help
Learn about a button or softkey	Press , then quickly press a button or softkey.
Learn about a menu item	Press , , or to display a feature menu. Highlight a menu item, then press ? twice quickly.
Get help using Help	Press 🕜 twice quickly. Select the help topic you need.

Understanding Feature Availability

Depending on your phone system configuration, features included in this Phone Guide might not be available to you or might work differently on your phone. Contact your support desk or system administrator for information about feature operation or availability.

You can access many features either by using a softkey or by pressing a line button. You can configure some features but your system administrator controls most of them.

Here are some details about using softkeys and line buttons to access features:

Feature	Softkey	Line Button Label and Icon
Call Back	CallBack	CallBack >
Call Forward	CFwdALL	Forward All
Call Park	Park	Park ⊳
Call PickUp	PickUp	PickUp >
Conference	Confrn	Conference 🚻
Conference List	ConfList	Conference List
Do Not Disturb	DND	Do Not Disturb O or Do Not Disturb O
End Call	EndCall	End Call >
Group Pickup	GPickUp	Group PickUp
Hold	Hold	Hold 🔘
Hunt Group	HLog	Hunt Group O or Hunt Group O
Malicious Call Identification	MCID	Malicious Call ID ⊳
Meet Me Conferencing	MeetMe	MeetMe >
Mobility	Mobility	Mobility 🗎
New Call	New Call	New Call >
Other PickUp	OPickUp	Other PickUp >
Quality Reporting Tool	QRT	Quality Reporting Tool
Redial	Redial	Redial >
Remove Last Conference Party	RmLstC	Remove Last Participant
Transfer	Transfer	Transfer 📞
Video Mode Command	VidMode	Video

Understanding SIP vs. SCCP

Your phone can be configured to work with one of two signaling protocols: SIP (Session Initiation Protocol) or SCCP (Skinny Call Control Protocol). Your system administrator determines this configuration.

Phone features can vary depending on the protocol. This Phone Guide indicates which features are protocol-specific. To learn which protocol your phone is using, you can ask your system administrator or you can choose > Model Information > Call Control Protocol on your phone.

Basic Call Handling

You can perform basic call-handling tasks using a range of features and services. Feature availability can vary; see your system administrator for more information.

Placing a Call—Basic Options

Here are some easy ways to place a call on your Cisco Unified IP Phone.

If you want to	Then	For more information, see
Place a call using the handset	Pick up the handset and enter a number.	An Overview of Your Phone, page 7
Place a call using the speakerphone	Press and enter a number.	Using a Handset, Headset, and Speakerphone, page 48
Place a call using a headset	Press and enter a number. Or if is lit, press New Call and enter a number.	Using a Handset, Headset, and Speakerphone, page 48
Redial a number	Press Redial to dial the last number, or press the Navigation button (with the phone idle) to see your Placed Calls.	Using Call Logs, page 53
Place a call when another call is active (using the same line)	 Press Hold. Press New Call. Enter a number. 	Using Hold and Resume, page 21
Dial from a call log	1. Choose > Missed Calls, Received Calls, or Placed Calls.	Using Call Logs, page 53
	2. Select the listing or scroll to it and go off-hook.	

Tips

- You can dial on-hook, without a dial tone (pre-dial). To pre-dial, enter a number, then go off-hook by lifting the handset or pressing Dial, , or .
- When you pre-dial, your phone tries to anticipate the number you are dialing by displaying
 matching numbers (if available) from your Placed Calls log. This is called Auto Dial. To call a
 number displayed with Auto Dial, press the number, or scroll to it and go off-hook.
- If you make a mistake while dialing, press << to erase digits.

- If parties on a call hear a beep tone, the call might be monitored or recorded. Ask your system administrator for more information.
- Your phone might be set up for international call logging, which is indicated by a "+" symbol on the call logs, redial, or call directory entries. See your system administrator for more information.

Placing a Call—Additional Options

You can place calls using special features and services that might be available on your phone. See your system administrator for more information about these additional options.

If you want to	Then	For more information, see
Place a call while another call is active (using a different	1. Press of for the new line. The first call is automatically placed on hold.	Using Hold and Resume, page 21
line)	2. Enter a number.	
Speed dial a number	Do one of the following:	Speed Dialing, page 34
	• Press (a speed-dial button).	
	• Use the Abbreviated Dial feature.	
	• Use the Fast Dial feature.	
Dial from a corporate directory on the phone	1. Choose > Corporate Directory (exact name can vary).	Using Call Logs, page 53
	2. Enter a name and press Search.	
	3. Highlight a listing and go off-hook.	
Dial from a corporate directory on your personal computer using	Open a web browser and go to a WebDialer-enabled corporate directory.	Using Cisco WebDialer, page 73
Cisco WebDialer	2. Click the number that you want to dial.	
Use CallBack to receive notification when a busy or	1. Press CallBack while listening to the busy tone or ring sound.	Your system administrator
ringing extension is available	2. Hang up. Your phone alerts you when the line is free.	
	3. Place the call again.	
See if a line associated with a speed-dial, call record, or directory listing is busy before placing a call to that line	Look for Busy Lamp Field indicators.	Using BLF to Determine a Line State, page 41

If you want to	Then	For more information, see
Make a priority (precedence) call (SCCP phones only)	Enter the MLPP access number, then enter the phone number.	Prioritizing Critical Calls, page 43
Dial from a Personal Address Book (PAB) entry	1. Choose > Personal Directory to log in.	Using Personal Directory on Your Phone, page 56
	2. Choose Personal Address Book and search for a listing.	
Place a call using a billing or	1. Dial a number.	Your system
tracking code (SCCP phones only)	2. After the tone, enter a client matter code (CMC) or a forced authorization code (FAC).	administrator
Place a call using your Cisco Extension Mobility profile Log in to the Cisco Extension Mobility service on a phone.		Using Cisco Extension Mobility, page 45
Make a call from a mobile phone using Mobile Voice	1. Dial your assigned Mobile Voice access number.	Using Cisco Extension Mobility, page 45
Access	2. Enter your mobile phone number (if requested) and PIN.	
	3. Press 1 to make a call to an enterprise IP phone.	
	4. Dial a desktop phone number other than your desktop phone number.	
	Note Before using Mobile Voice Access to make a call, obtain your Mobile Voice Access number and End user PIN from your system administrator.	

If you want to	Then	For more information, see
Place a call using Fast Dials.	Note Before using this option, your system administrator must configure this feature and assign a service URL to the line button. Contact your system administrator for more information. 1. Press the Fast Dial line button. 2. Scroll to or press the index number	Configuring Fast Dials on the Web, page 63 Using Your Personal Address Book on the Web, page 61
	to find the entry. 1. Press Dial.	
Place a call using Personal Address Book (PAB).	Note Before using this option, your system administrator must configure this feature and assign a service URL to the line button. Contact your system administrator for more information.	Using Your Personal Address Book on the Web, page 61 Configuring Fast Dials on the Web, page 63
	 Press the PAB line button. Access the contact and select the number. 	
	3. Press Dial.	

Answering a Call

You can answer a call by lifting the handset, or you can use other options if they are available on your phone.

If you want to	Then	For more information, see
Answer with a wired headset	Press , if unlit. Or, if is already lit, press Answer or (flashing).	Using a Handset, Headset, and Speakerphone, page 48
	If you are using a wireless headset, refer to the wireless headset documentation.	
Answer with the speakerphone	Press , Answer, or (flashing).	Using a Handset, Headset, and Speakerphone, page 48
Switch from a connected call to answer a new call	Press Answer , or if the call is ringing on a different line, press (flashing).	Using Hold and Resume, page 21

If you want to	Then	For more information, see
Answer using call waiting	Press Answer.	Using Hold and Resume, page 21
Send a call to a voice message system	Press iDivert.	Sending a Call to a Voice Message System, page 25
Auto-connect calls	Use AutoAnswer.	Using AutoAnswer with a Headset or Speakerphone, page 50
Retrieve a parked call on another phone	Use Call Park or Directed Call Park.	Storing and Retrieving Parked Calls, page 36
Use your phone to answer a call ringing elsewhere	Use Call Pickup.	Picking Up a Redirected Call on Your Phone, page 35
Answer a priority call (SCCP phones only)	Hang up the current call and press Answer.	Prioritizing Critical Calls, page 43
Answer a call on your mobile phone or other remote destination	 Set up Mobile Connect and answer your phone. When you enable Mobile Connect: Your desktop and remote destinations receive calls simultaneously. When you answer the call on your desktop phone, the remote destinations stop ringing, are disconnected, and display a missed call message. When you answer the call on one remote destination, the other remote destinations stop ringing, are disconnected, and a missed call message is shown on the other remote destinations. 	Using Cisco Extension Mobility, page 45

Tip

If parties on a call hear a beep tone, the call might be monitored or recorded. Ask your system administrator for more information.

Ending a Call

To end a call, hang up. Here are some more details.

If you want to	Then
Hang up while using the handset	Return the handset to its cradle. Or press EndCall.
Hang up while using a wired headset	Press . If you want to keep headset mode activate, press EndCall.
	If you are using a wireless headset, refer to the wireless headset documentation.
Hang up while using the speakerphone	Press of or EndCall.
Hang up one call but preserve another call on the same line	Press EndCall. If necessary, remove the call from hold first.

Using Hold and Resume

If the Hold Reversion feature is enabled for your phone, a call that you put on hold reverts back to ringing after a certain period of time. The 'reverting' call remains on hold until you resume it or until Hold Reversion times out.

Your phone indicates the presence of a reverting call by:

- Alerting you at intervals with a single ring (or flash or beep, depending on your phone line setting).
- Briefly displaying a "Hold Reversion" message in the status bar at the bottom of the phone screen.
- Displaying the animated Hold Reversion icon next to the caller ID for the held call.
- Displaying a flashing amber line button (depending on the line state).

If you want to	Then
Put a call on hold	1. Make sure the call you want to put on hold is highlighted.
	2. Press Hold.

If you want to	Then
Remove a call from hold on the current line	1. Make sure that the appropriate call is highlighted.
	2. Press Resume.
Remove a call from hold on a different line	1. Press the appropriate line button: or (flashing). Doing so might cause a held call to resume automatically:
	 If there is a reverting call on the line, that call will resume.
	 If there is more than one reverting call on the line, the oldest reverting call will resume.
	 If a non-reverting held call is the only call on the line, it will resume.
	2. If necessary, scroll to the appropriate call and press Resume.

Tips

- Engaging the Hold feature typically generates music or a beeping tone.
- If you receive an alert for an incoming call and a reverting call at the same time, by default your phone will shift the focus of the phone screen to display the incoming call. Your system administrator can change this focus priority setting.
- If you use a shared line, Hold Reversion rings only on the phone that put the call on hold, not on the other phones that share the line.
- The duration between Hold Reversion alerts is determined by your system administrator.

Using Mute

With Mute enabled, you can hear other parties on a call but they cannot hear you. You can use Mute with the handset, speakerphone, or a headset.

If you want to	Then
Toggle Microphone on	Press .
Toggle Microphone off	Press .

Switching Between Multiple Calls

You can switch between multiple calls on one or more lines. If the call that you want to switch to is not automatically highlighted, use the Navigation button to scroll to it.

If you want to	Then
Switch between connected calls on one line	 Make sure the call that you want to switch to is highlighted. Press Resume. Any active call is placed on hold and the selected call is resumed.
Switch between connected calls on different lines	Press of for the line that you are switching to. If a single call is holding on the line, the call automatically resumes. If multiple calls are holding, highlight the appropriate call and press Resume.
Switch from a connected call to answer a ringing call	Press Answer, or if the call is ringing on a different line press (flashing). Any active call is placed on hold and the selected call is resumed.

Switching an In-Progress Call to Another Phone

You can switch in-progress calls between the desktop phone and your mobile phone or other remote destination.

If you want to	Then
Switch an in-progress call on your desktop phone to a mobile phone	 Press the Mobility softkey and select Send call to mobile. Answer the in-progress call on your mobile phone. The desktop phone line button turns red and handset icons and the calling party number appear on the phone display. You cannot use same phone line for any other calls, but if your desk phone supports multiple lines, you can use another line to make or receive calls.
Switch an in-progress call from a mobile phone to your desktop phone	 Hang up the call on your mobile phone to disconnect the mobile phone, but not the call. Press Resume on your desk phone within 4 seconds and start talking on the desk phone.

Viewing Multiple Calls

Understanding how multiple calls are displayed on your phone can help you organize your call-handling efforts.

In standard viewing mode, your phone displays calls as follows for the highlighted line:

- Calls with the highest precedence and longest duration display at the top of the list.
- Calls of a similar type are grouped together. For example, calls that you have interacted with are grouped near the top, and calls on hold are grouped last.

You can use these additional methods to view multiple calls on multiple lines:

If you want to	Then
View calls on another line	 Press . Immediately press the line button .
Switch to call overview mode	Press of for the highlighted line. The phone switches to call overview mode, displaying one call per line. The displayed call is either the active call or the held call with the longest duration. To return to standard viewing mode, press , then immediately press the line button.

Transferring Calls

Transfer redirects a connected call. The target is the number to which you want to transfer the call.

If you want to	Then
Transfer a call without talking to the transfer recipient	 From an active call, press Transfer. Enter the target number.
	3. Press Transfer again to complete the transfer or EndCall to cancel.
	Note If your phone has on-hook transfer enabled, complete the transfer by hanging up.
Talk to the transfer recipient before transferring a call (consult transfer)	1. From an active call, press Transfer.
	2. Enter the target number.
	3. Wait for the transfer recipient to answer.
	4. Press Transfer again to complete the transfer or EndCall to cancel.
	Note If your phone has on-hook transfer enabled, complete the transfer by hanging up.

If you want to	Then
Transfer two current calls to each other (direct transfer) without staying on the line (SCCP phones only)	 Scroll to highlight any call on the line. Press Select.
	3. Repeat this process for the second call.
	4. With one of the selected calls highlighted, press DirTrfr. (To display DirTrfr, you might need to press more.)
	The two calls connect to each other and drop you from the call.
	Note If you want to stay on the line with the callers, use Join instead.
Redirect a call to a voice message system	Press iDivert. For more information, see Sending a Call to a Voice Message System, page 25.

Tips

- If your phone has on-hook transfer enabled, complete the call by hanging up.
- If on-hook transfer is *not* enabled on your phone, hanging up without pressing **Transfer** again places the call on hold.
- You cannot use **Transfer** to redirect a call on hold. Press **Resume** to remove the call from hold before transferring it.

Sending a Call to a Voice Message System

You can use **iDivert** to send an active, ringing, or on-hold call to your voice message system. Depending on the type of call and your phone configuration, you can also use **iDivert** to send the call to another party's voice message system.

- If the call was originally sent to someone else's phone, iDivert allows you to redirect the call either to your own voice message system or to the original called party's voice message system. Your system administrator must make this option available to you.
- If the call was sent to you directly (not transferred or forwarded to you), or if your phone does not support the option described above, using iDivert redirects the call to your voice message system.

If you want to	Then
Send an active, ringing, or on-hold call	Press iDivert. One of two things occurs:
to a voice message system	The call is transferred to your voice message system.
	• Your phone screen displays a menu that allows you to choose between your voice message system or the voice message system of the original called party. Choose an option to redirect the call.

Tip

If your phone displays a menu that disappears before you make your selection, you can press iDivert again to re-display the menu. You can also ask your system administrator to configure a longer timeout value.

Forwarding Calls to Another Number

You can use the call forwarding features to redirect incoming calls from your phone to another number.

Your system administrator might allow you to choose from two types of call forwarding features:

- Unconditional call forwarding (Call Forward All)—Applies to all calls that you receive.
- Conditional call forwarding (Call Forward No Answer, Call Forward Busy, Call Forward No Coverage)—Applies to certain calls that you receive, according to conditions.

You can access Call Forward All on your phone or from your User Options web pages; conditional call forwarding features are accessible only from your User Options web pages. Your system administrator determines which call forwarding features are available to you.

If you want to	Then
Set up Call Forward All on your primary line	Press CFwdALL or Forward All and enter a target phone number.
Cancel Call Forward All on your primary line	Press CFwdALL or Forward All.
Verify that Call Forward All is	Look for:
enabled on your primary line	• The call forward icon above your primary phone number: 🙀.
	The call forwarding target number in the status line.
Set up or cancel call forwarding remotely, or for a	1. Log in to your User Options web pages. (See Accessing Your User Options Web Pages, page 60.)
non-primary line	2. Access your call forwarding settings (See Controlling Line Settings on the Web, page 68.)
	Note When call forwarding is enabled for any line other than the primary line, your phone does not provide you with any confirmation that calls are being forwarded. Instead, you must confirm your settings in the User Options web pages.

Tips

• Enter the call forward target number exactly as you would dial it from your phone. For example, enter an access code or the area code, if necessary.

- You can forward your calls to a traditional analog phone or to another IP phone, although your system administrator might restrict the call forwarding feature to numbers within your company.
- Call forwarding is phone line specific. If a call reaches you on a line where call forwarding is not enabled, the call will ring as usual.
- Your system administrator can enable a call forward override feature that allows the person receiving your forwarded calls to reach you. With override enabled, a call placed from the target phone to your phone is not forwarded, but rings through.
- Your phone may reject your attempt to set up Call Forward All directly on the phone if the target number that you enter would create a call forwarding loop or would exceed the maximum number of links permitted in a call forwarding chain.

Using Do Not Disturb

You can use the Do Not Disturb (DND) feature to turn off only the ringer on your phone or to turn off all audible and visual notifications of incoming calls. Your system administrator enables DND for your phone.

When DND and Call Forward All are both enabled on your phone, calls are forwarded and the caller does not hear a busy tone.

DND interaction with other types of calls includes:

- DND does not affect intercom calls or non-intercom priority calls.
- If both DND and auto-answer are enabled, only intercom calls will be auto-answered.

If you want to	Then
Turn on DND	Press DND or Do Not Disturb .
	"Do Not Disturb" displays on the phone, the DND lights, and the ring tone is turned off.

Turn off DND	Press DND or Do Not Disturb .
Customize DND settings	If your system administrator configured DND settings to appear on the User Options page, follow these steps:
	1. Log in to your User Options web pages. See Accessing Your User Options Web Pages, page 60.
	2. From the drop-down menu, choose User Options > Device.
	3. Set the following options:
	 Do Not Disturb—Set to enable/disable DND.
	 DND Option—Choose either Call Reject (to turn off all audible and visual notifications) or Ringer Off (to turn off only the ringer).
	 DND Incoming Call Alert (applies to either DND option set)—Set the alert to beep only, flash only, disable the alert, or choose "None" (to use the "Alert" setting configured by your system administrator).

Tips

- When DND is active, DND blocking is applied to all the lines on your phone.
- When DND and Call Forward All are both enabled on your phone, Call Forward All takes precedence on incoming calls. That is, calls will be forwarded and the caller will not hear a busy tone.
- Intercom and Multilevel Precedence and Preemption (MLPP), park reversion and hold reversion all override DND.
- Your system administrator enables the DND feature on your phone.

Making Conference Calls

Your Cisco Unified IP Phone allows you to talk simultaneously with multiple parties in a conference call.

Using Conference Features

You can create a conference in various ways, depending on your needs and the features that are available on your phone.

- Conference—Allows you to create a standard (ad hoc) conference by calling each participant. Use the **Confrn** softkey or the **Conference** button. Conference is available on most phones.
- Join—Allows you to create a standard (ad hoc) conference by combining existing calls. Use the **Join** softkey or button.

- cBarge—Allows you to create a standard (ad hoc) conference by adding yourself to a call on a shared line. Press a line button or use the **cBarge** softkey or button. cBarge is available only on phones that use shared lines.
- Meet-Me—Allows you to create or join a conference by calling a conference number. Use the MeetMe softkey or button.

Using Conference

Conference allows you to call each participant. Conference is available on most phones.

If you want to	Then
Create a conference	1. From a connected call, press Confrn or Conference. (You may need to press the more softkey to see Confrn.)
	2. Enter the participant's phone number.
	3. Wait for the call to connect.
	4. Press Confrn or Conference again to add the participant to your call.
	5. Repeat to add additional participants.
Add new participants to an	Repeat the steps listed above.
existing conference	Your system administrator determines whether non-initiators of a conference can add or remove participants.
See a list of participants or remove participants	See Viewing or Removing Conference Participants, page 32.

Using Join

Join allows you to combine two or more existing calls to create a conference in which you are a participant.

If you want to	Then
Create a conference by joining together existing calls that are	1. From an active call, highlight another call that you want to include in the conference and press Select .
on a single phone line	Selected calls display this icon 🗹.
	2. Repeat this step for each call that you want to add.
	3. Press Join. (You may need to press the more softkey to see Join.)
Create a conference by joining together existing calls that are on multiple phone lines	1. From an active call, press Join. (You may need to press the more softkey to see Join.)
	2. Press the green flashing line button of for the call(s) that you want to include in the conference.
	One of the following occurs:
	The calls are joined.
	• A window opens on your phone screen prompting you to select the call(s) that you want to join. Highlight the call(s) and press Select , then press Join to complete the action.
	Note If your phone does not support Join for calls on multiple lines, transfer the calls to a single line before using Join.
See a list of participants or remove participants	Press ConfList or Conference List. (You may need to press the more softkey first.) See Viewing or Removing Conference Participants, page 32.

Tips

- If you frequently join more than two parties into a single conference, you might find it useful to first select the calls that you want to join, then press **Join** to complete the action.
- When Join completes, caller ID changes to "Conference."
- You may be able to combine multiple conference calls by using the **Join** or **DirTrfr** softkeys. Check with your system administrator to see whether this feature is available to you.

Using cBarge

You can create a conference by using cBarge to add yourself to a call on a shared line.

If you want to	Then
Create a conference by barging a call on a shared line	Press the line button for the shared line. In some cases, you must highlight the call and press cBarge to complete the action.
	See Using Barge to Add Yourself to a Shared-Line Call, page 39 for more information.
See a list of participants or remove participants	See Viewing or Removing Conference Participants, page 32.

Using Meet-Me

Meet-Me conferencing allows you to start or join a conference by calling the conference number.

If you want to	Then
Start a Meet-Me conference	1. Obtain a Meet-Me phone number from your system administrator.
	2. Distribute the number to participants.
	3. When you are ready to start the meeting, go off-hook to get a dial tone, then press MeetMe .
	4. Dial the Meet-Me conference number.
	Participants can now join the conference by dialing in.
	Note Participants hear a busy tone if they call the conference before the initiator has joined. In this case, participants must call back.
Join a Meet-Me conference	Dial the Meet-Me conference number (provided by the conference initiator).
	Note You will hear a busy tone if you call the conference before the initiator has joined. In this case, try your call again.
End a Meet-Me conference	All participants must hang up.
	The conference does not automatically end when the conference initiator disconnects.

Tip

If you call a secure Meet-Me conference number from a non-secure phone, your phone displays the message, "Device Not Authorized." For more information, see Making and Receiving Secure Calls, page 42.

Viewing or Removing Conference Participants

During a standard (ad hoc) conference, you can view a list of participants and remove participants.

View a list of conference	Press ConfList or Conference List.
participants	Participants are listed in the order in which they join the conference with the most recent additions at the top.
Get an updated list of conference participants	While viewing the conference list, press Update.
See who initiated the conference	While viewing the conference list, locate the participant listed at the bottom of the list with an asterisk (*) next to the name.
Remove any conference participant	While viewing the conference list, highlight the participant's name and press Remove.
Drop the last participant added to the conference	While viewing the conference list, press RMLstC or Remove Last Participant.
Verify that a conference call is secure	Look for the for or for icon after "Conference" on the phone screen.
Verify that a participant is calling from a secure phone	Look for the or icon beside the participant's name on the phone screen.
Add more participants	See Using Conference, page 29.

Placing or Receiving Intercom Calls

You can make an intercom call to a target phone that auto-answers the call in speakerphone mode with mute activated. The one-way intercom call allows you to deliver a short message to the recipient. If the recipient's handset or headset is in use, the audio is sent to the device that is use. Any current call activity that your recipient is engaged in continues simultaneously.

The target destination receives an intercom-alert tone and can then choose to:

- Listen to the caller with your microphone muted (you can hear the caller but the caller cannot hear you).
- End the intercom call by pressing the **EndCall** softkey with the intercom call in focus. Do this if you do not want to hear the message.
- Talk to the caller by pressing the active intercom button, and use either the handset, headset or speaker. The intercom call becomes a two-way connection so that you can converse with the caller.

When using the intercom feature, be aware of the following:

- From an intercom line, you can only dial other intercom lines.
- You can use only one intercom line at a time.
- When your active call is being monitored or recorded, you cannot receive or place intercom calls.

• You cannot place an intercom call on hold.



If you log into your desk phone every day using your Extension Mobility profile, make sure that your system administrator configures your Extension Mobility profile to include the intercom feature.

If you want to	Then
Place an intercom call to a preconfigured intercom target	Press (intercom target line) and, after you hear the intercom-alert tone, begin speaking.
Place an intercom call to any intercom number	Press (an intercom line). Enter the intercom target number or press a speed-dial number for your target. After you hear the intercom-alert tone, begin speaking.
Receive an intercom call	When you hear the intercom-alert tone, handle the call in one of these ways: • Listen to the message in one-way audio. • Speak to the caller by pressing (active intercom line). • Press EndCall with the intercom call in focus.

Advanced Call Handling

Advanced call-handling tasks involve special features that your system administrator might configure for your phone depending on your call-handling needs and work environment.

Speed Dialing

Speed dialing allows you to enter an index number, press a button, or select a phone screen item to place a call. Depending on configuration, your phone can support several speed-dial features:

- Speed-dial buttons
- Abbreviated Dialing
- Fast Dials



- To set up speed-dial buttons and Abbreviated Dial, you must access your User Options web pages. See Accessing Your User Options Web Pages, page 60.
- To set up Fast Dials, you must access the Personal Directory feature. See Using Personal Directory on Your Phone, page 56.
- Alternately, your system administrator can configure speed-dial features for you.

If you want to	Then
Use speed-dial	1. Set up speed-dial buttons. See Setting Up Speed Dials on the Web, page 64.
buttons	2. To place a call, press (a speed-dial button).
	Note If your phone supports the Busy Lamp Field (BLF) speed-dial feature, you can see if the speed-dial number is busy before dialing. See Using BLF to Determine a Line State, page 41.
Use Abbreviated Dial	1. Set up Abbreviated Dialing codes. See Setting Up Speed Dials on the Web, page 64.
	2. To place a call, enter the Abbreviated Dialing code and press AbbrDial.
Use Fast Dial	1. Create a Personal Address B ook entry and assign a Fast Dials code. See Using Your Personal Address Book on the Web, page 61.
	2. To place a call, access the Fast Dial service on your phone. See Using Personal Directory on Your Phone, page 56.

Picking Up a Redirected Call on Your Phone

Call PickUp allows you to answer a call that is ringing on a coworker's phone by redirecting the call to your phone. You might use Call PickUp if you share call-handling with coworkers.

If you want to	Then
Answer a call that is ringing on another extension within your call	1. Press the PickUp softkey or button. (You might have to go off-hook to display the softkey.)
pickup group	If your phone supports auto-pickup, you are now connected to the call.
	2. If the call rings, press Answer to connect to the call.
Answer a call that is ringing on another extension outside of your	1. Press the GPickUp softkey or the Group PickUp button. (You might have to go off-hook to display the softkey.)
call pickup group	2. Enter the group pickup number.
	If your phone supports auto-pickup, you are now connected to the call.
	3. If the call rings, press Answer to connect to the call.
Answer a call that is ringing on another extension in your group or	1. Press the OPickUp softkey or the Other PickUp button. (You might have to go off-hook to display the softkey.)
in an associated group	If your phone supports auto-pickup, you are now connected to the call.
	2. If the call rings, press Answer to connect to the call.
Answer a call that is ringing on a particular extension (line number)	1. Press the GPickUp softkey or the Group PickUp button. (You might have to go off-hook to display the softkey.)
	2. Enter the line number with the call that you want to pick up. For example, if the call that you want to pick up is ringing on line 12345, enter 12345.
	If your phone supports auto-pickup, you are now connected to the call.
	3. If the call rings, press Answer to connect to the call.

Tips

- If multiple calls are available for pick up, your phone picks up the oldest call first (the call that has been ringing for the longest time).
- If you press GPickUp or Group PickUp and enter a line number, your phone picks up the ringing call on that particular line (if available).

• If you have multiple lines and want to pick up the call on a non-primary line, first press (a) for the desired line, then press a Call PickUp softkey or button.



- Depending on how your phone is configured, you might receive an audio and/or visual alert about a call to your pickup group.
- If you use the BLF Pickup feature on your phone, see Using BLF to Determine a Line State, page 41.

Storing and Retrieving Parked Calls

You can park a call when you want to store the call so that you or someone else can retrieve it from another phone in the Cisco Unified Communications Manager system (for example, a phone at a coworker's desk or in a conference room).

You can park a call by using these methods:

- Call Park—Use the Park softkey to store the call. Your phone displays the call park number where the system stored your call. You must record this number and then use the same number to retrieve the call.
- Directed Call Park—Use the Transfer softkey to direct the call to an available directed call park number that you dial or speed dial. To retrieve the call from a directed call park number, dial the park retrieval prefix, then dial or speed dial the same directed call park number. You can use a Directed Call Park button to speed dial the directed call park number and to monitor whether a directed call park number is occupied or available.

If you want to	Then
Store an active call using Call Park	1. During a call, press Park. (You may need to press the more softkey to see Park.)
	2. Note the call park number displayed on your phone screen.
	3. Hang up.
Retrieve a parked call	Enter the call park number from any Cisco Unified IP Phone in your network to connect to the call.

If you want to	Then
Direct and store an active call at a directed call park number	 During a call, press Transfer. Press the Directed Call Park with the park-unoccupied icon to speed dial the directed call park number.
	A Directed Call Park (flashing) with park-occupied icon indicates the directed call park number is not available.
	3. Press Transfer again to finish storing the call.
Retrieve a parked call from a directed call park number	From any Cisco Unified IP Phone in your network, enter the park retrieval prefix and dial the directed call park number. Or after entering the park retrieval prefix, press the (flashing) with to connect to the call.

- You have a limited time to retrieve a parked call before it reverts to ringing at the original number. See your system administrator for details.
- Your system administrator can assign Directed Call Park buttons to available line buttons on your Cisco Unified IP Phone 7962G that has a Cisco Unified IP Phone Expansion Module (the Cisco Unified IP Phone 7942G does not support the Cisco Unified IP Phone Expansion Module).
- You can dial directed call park numbers if you do not have Directed Call Park buttons. However, you will not be able to see the status of the directed call park number.

Logging Out of Hunt Groups

If your organization receives a large number of incoming calls, you might be a member of a hunt group. A hunt group includes a series of directory numbers that share the incoming call load. When the first directory number in the hunt group is busy, the system hunts for the next available directory number in the group and directs the call to that phone.

When you are away from your phone, you can prevent hunt group calls from ringing your phone by logging out of hunt groups.

If you want to	Then
Log out of hunt groups to temporarily block hunt group calls	Press HLog or Hunt Group . Your phone screen displays, "Logged out of Hunt Group."
Log in to receive hunt group calls	Press HLog or Hunt Group . When logged in, the Hunt Group button is lit.

Tip

Logging out of hunt groups does not prevent non-hunt group calls from ringing your phone.

Using a Shared Line

Your system administrator might ask you to use a shared line if you:

- Have multiple phones and want one phone number
- Share call-handling tasks with coworkers
- Handle calls on behalf of a manager

Understanding Shared Lines

Remote-in-Use Icon

The Remote-in-Use icon appears when another phone that shares your line has a connected call. You can place and receive calls as usual on the shared line, even when the Remote-in-Use icon appears.

Sharing Call Information and Barging

Phones that share a line each display information about calls that are placed and received on the shared line. This information might include caller ID and call duration. (See the Privacy section for exceptions.)

When call information is visible in this way, you and coworkers who share a line can add yourselves to calls using either Barge or cBarge. See Using Barge to Add Yourself to a Shared-Line Call, page 39.

Privacy

If you do not want coworkers who share your line to see information about your calls, enable the Privacy feature. Doing so also prevents coworkers from barging your calls. See Preventing Others from Viewing or Barging a Shared-Line Call, page 40.



The maximum number of calls that a shared line supports can vary by phone.

Using Barge to Add Yourself to a Shared-Line Call

You can use barge features (cBarge or Barge) to add yourself to calls on your shared line. (Calls must be non-private calls. See Understanding Shared Lines, page 38.)

Understanding Barge Features

cBarge and Barge

Depending on how your phone is configured, you can add yourself to a non-private call on a shared line using either cBarge or Barge:

- cBarge converts the call into a standard conference, allowing you to add new participants. (See Making Conference Calls, page 28 for information about standard conferences.)
- Barge allows you to add yourself to the call but does not convert the call into a conference or allow you to add new participants.

Single-button and Multi-touch Barge

Your system administrator determines whether the barge feature on your phone (cBarge or Barge) operates as a single-button or multi-touch feature.

- Single button barge allows you to press a line button to barge a call (if only one call is on the line).
- Multi-touch barge allows you to view call information before barging.

Using Barge Features

The table below describes how to use barge features (cBarge or Barge) to add yourself to a call on a shared line.

If you want to	Then
See if the shared line is in use	Look for the remote-in-use icon property next to a red line button.
Add yourself to a call on a shared line	Press the red line button of for the shared line.
	One of the following occurs:
	You are added to the call.
	• A window opens on your phone screen prompting you to select the call that you want to barge. Press Barge or cBarge to complete the action. (You may need to press the more softkey to display Barge or cBarge .)
View conference participants (if you used cBarge)	See Viewing or Removing Conference Participants, page 32.

- When you barge a call, other parties hear a beep tone announcing your presence. With cBarge, other parties hear a brief audio interruption and the phone screen changes to display conference details.
- If a phone that is using the shared line has Privacy enabled, call information and barge softkeys will not appear on the other phones that share the line.
- You will be disconnected from a call that you have joined using Barge if the call is put on hold, transferred, or turned into a conference call.

Preventing Others from Viewing or Barging a Shared-Line Call

If you share a phone line, you can use the Privacy feature to prevent others who share the line from viewing or barging (adding themselves to) your calls.

If you want to	Then
Prevent others from viewing or barging calls on a shared line	 Press Private To verify that Privacy is on, look for the Privacy-enabled icon next to an amber line button
Allow others to view or barge calls on a shared line	 Press Private . To verify that Privacy is off, look for the Privacy-disabled icon next to an unlit line button.

Tips

- If the phone that shares your line has Privacy enabled, you can make and receive calls using the shared line as usual.
- The Privacy feature applies to all shared lines on your phone. Consequently, if you have multiple shared lines and Privacy is enabled, coworkers will not be able to view or barge calls on any of your shared lines.

Using BLF to Determine a Line State

Busy Lamp Field features (BLF) allow you to view the state of a phone line that is associated with a speed-dial button, call log, or directory listing on your phone. If you use BLF Pickup, you can answer a ringing call for the line that you are monitoring. Your system administrator determines which BLF features are configured for your phone.

If you want to	Then
See the state of a line listed in a call log or directory	Look for one of these indicators next to the line number: Line is in-use. Line is idle. BLF indicator unavailable for this line. Line is in Do Not Disturb state.
See the state of a speed-dial line	Look for one of these indicators next to the line number: + Line is in-use. + Line is idle. BLF indicator unavailable for this line. + Line is in Do Not Disturb state. (flashing)—Line is ringing (BLF Pickup only).
Use BLF Pickup to answer a call ringing on a coworker's phone	Press the BLF Pickup button while the line is ringing. The call is redirected to the next available line on your phone. (If you want to specify a line, first press a line button, then press the BLF button.) If your phone supports auto-pickup, the call connects automatically. Otherwise, the call rings on your phone for you to answer. Note If you press the BLF Pickup button when the monitored line is not ringing, your phone will speed dial the line number.

Tips

- Your phone might play an audible indicator to alert you when a call is ringing on the monitored line (BLF Pickup only).
- BLF Pickup answers the oldest ringing call first (if the line that you are monitoring has more than one ringing call).

Making and Receiving Secure Calls

Depending on how your system administrator has configured your phone system, your phone might support making and receiving secure calls.

Your phone is capable of supporting these types of calls:

- Authenticated call—The identities of all phones participating in the call have been verified.
- *Encrypted* call—The phone is receiving and transmitting encrypted audio (your conversation) within the Cisco IP network. Encrypted calls are also authenticated.
- Protected call—The phone is receiving and transmitting encrypted audio (your conversation) with another phone. A protected call provides a similar level of security as an encrypted call, but adds additional security. If the call is protected at both ends, a security tone plays at the beginning of the conversation. Some features, such as conference calling, shared lines, Extension Mobility, and join across lines are not available when protected calling is configured. Protected calls are not authenticated.
- *Non-secure* call—At least one of the participating phones or the connection does not support these security features, or the phones cannot be verified.

If you want to	Then
Check the security level of a call or conference	Look for a security icon in the top right corner of the call activity area, next to the call duration timer:
	Authenticated call or conference
	Encrypted call or conference
	Non-secure call or conference
Verify that the phone on the other end of the connection is also secure	Look for the encrypted lock icon and listen for a security tone at the beginning of the call.
	When encrypted calling is configured on your phone, the lock icon is displayed. However, the conversation is protected only when the security tone plays at the beginning of a call, indicating that the call is protected on both ends of the call. If your call is connected to a non-protected phone, the secure tone is not played.
Determine if secure calls can be made in your company	Contact your system administrator.



There are interactions, restrictions, and limitations that affect how security features work on your phone. For more information, ask your system administrator.

Tracing Suspicious Calls

If you are receiving suspicious or malicious calls, your system administrator can add the Malicious Call Identification (MCID) feature to your phone. This feature enables you to identify an active call as suspicious, which initiates a series of automated tracking and notification messages.

If you want to	Then
Notify your system administrator about a suspicious or harassing call	Press MCID or Malicious Call ID. Your phone plays a tone and displays the message, "MCID successful".

Prioritizing Critical Calls

(SCCP phones only.)

In some specialized environments, such as military or government offices, you might need to make and receive urgent or critical calls. If you have the need for this specialized call handling, your system administrator can add Multilevel Precedence and Preemption (MLPP) to your phone.

Keep these terms in mind:

- Precedence indicates the priority associated with a call.
- *Preemption* is the process of ending an existing, lower priority call while accepting a higher priority call that is sent to your phone.

If you	Then
Want to choose a priority (precedence) level for an outgoing call	Contact your system administrator for a list of corresponding precedence numbers for calls.
Want to make a priority (precedence) call	Enter the MLPP access number (provided by your system administrator) followed by the phone number.
Hear a special ring (faster than usual) or special call waiting tone	You are receiving a priority (precedence) call. An MLPP icon on your phone screen indicates the priority level of the call.

If you	Then
Want to view priority level of a call	Look for an MLPP icon on your phone screen:
	Priority call
	Medium priority (immediate) call
	High priority (flash) call
	Highest priority (flash override) or Executive Override call
	Higher priority calls are displayed at the top of your call list. If you do not see an MLPP icon, the priority level of the call is normal (routine).
Want to accept a higher-priority call	Answer the call as usual. If necessary, end an active call first.
Hear a continuous tone interrupting your call	You or the other party are receiving a call that must preempt the current call. Hang up immediately to allow the higher priority call to ring through.

- When you make or receive an MLPP-enabled call, you will hear special ring tones and call waiting tones that differ from the standard tones.
- If you enter an invalid MLPP access number, a verbal announcement will alert you of the error.
- An MLPP-enabled call retains its priority and preemptive status when you:
 - Put the call on hold
 - Transfer the call
 - Add the call to a three-way conference
 - Answer the call using PickUp
- MLPP overrides the Do Not Disturb (DND) feature.

Using Cisco Extension Mobility

Cisco Extension Mobility (EM) allows you to temporarily configure a Cisco Unified IP Phone as your own. Once you log in to EM, the phone adopts your user profile, including your phone lines, features, established services, and web-based settings. Your system administrator must configure EM for you.

If you want to	Then
Log in to EM	1. Choose > EM Service (name can vary).
	2. Enter your user ID and PIN (provided by your system administrator).
	3. If prompted, select a device profile.
Log out of EM	1. Choose > EM Service (name can vary).
	2. When prompted to log out, press Yes.

Tips

- EM automatically logs you out after a certain amount of time. Your system administrator establishes this time limit.
- Changes that you make to your EM profile from your User Options web pages take effect immediately if you are logged in to EM on the phone; otherwise, changes take effect the next time you log in.
- Changes that you make to the phone from your User Options web pages take effect immediately if you are logged out of EM; otherwise, changes take effect after you log out.
- Local settings controlled by the phone are not maintained in your EM profile.

Managing Business Calls Using a Single Phone Number

With Mobile Connect and Mobile Voice Access installed, you can use your mobile phone to handle calls associated with your desktop phone number.

If you want to	Then
Configure Mobile Connect	Use the User Options web pages to set up remote destinations and create access lists to allow or block calls from specific phone numbers from being passed to the remote destinations. See Setting Up Phones and Access Lists for Mobile Connect, page 70.
Answer a call using your mobile phone	See Answering a Call, page 19.
Switch an in-progress call between your desk phone and a mobile phone	See Switching an In-Progress Call to Another Phone, page 23.

If you want to	Then
Put a call that has been picked up on your mobile phone on hold	1. Press the <i>Enterprise Hold</i> (name may vary) softkey.
	The other party is placed on hold.
	2. On your mobile phone, press the <i>Resume</i> (name may vary) softkey. See Switching an In-Progress Call to Another Phone, page 23.
Connect to Mobile Voice	1. From any phone, dial your assigned Mobile Voice Access number.
Access	2. Enter the number you are calling from, if prompted, and your PIN.
Turn on Mobile Connect	1. Dial your assigned Mobile Voice access number.
from your mobile phone	2. Enter your mobile phone number (if requested) and PIN.
	3. Press 2 to activate Mobile Connect.
	4. Choose whether to turn Mobile Connect on for all configured phones or just one:
	- All phones—Enter 2.
	 One phone—Enter 1 and then the remote destination you want to add, followed by #.
Make a call from your mobile phone	See Placing a Call—Additional Options, page 17.
Turn off Mobile Connect	1. Dial your assigned Mobile Voice Access number.
from your mobile phone	2. Enter your mobile phone number (if requested) and PIN.
	3. Press 3 to disable Mobile Connect.
	4. Choose whether to turn off Mobile Connect for all configured phones or just one:
	- All phones—Enter 2.
	 One phone—Enter 1 and then the remote destination you want to add, followed by #.
Turn on or off Mobile Connect access to all your	1. Press Mobility to display the current remote destination status (Enabled or Disabled).
remote destinations from your desk phone	2. Press Select to change the status.
your desk phone	3. Press Exit.

- When calling Mobile Voice Access, you must enter the number you are calling and your PIN if any of the following is true:
 - The number you are calling from is not one of your remote destinations.
 - The number is blocked by you or your carrier (shown as "Unknown Number").

- The number is not accurately matched in the Cisco Unified Communications Manager database; for example, if your number is 510-666-9999, but it is listed as 666-9999 in the database, or your number is 408-999-6666, but it is entered as 1-408-999-6666 in the database.
- If you incorrectly enter any requested information (such as mobile phone number or PIN) three times in a row, the Mobile Voice Access call disconnects, and you are locked out for a period of time. Contact your system administrator if you need assistance.

Using a Handset, Headset, and Speakerphone

You can use your phone with these audio devices: a handset, headset, or speakerphone.

The phone is off-book when the handset is lifted or another audio device is in use.

The phone is *on-hook* when the handset is in its cradle and other audio devices are not in use.

Using a Handset

If you want to	Then
Go off-hook	Lift the handset.
Go on-hook	Replace the handset in the cradle.
Switch to the speakerphone or headset during a call	Press or or, then hang up the handset.
Adjust the volume level for a call	Press Save to preserve the volume level for future calls.

Using a Headset

Your phone supports four- or six-wire headset jacks for wired headsets, and also supports wireless headsets. For information about purchasing headsets, see Using a Headset, page 5.

However, if you use a wireless headset, refer to the wireless headset documentation for information.

If you want to	Then
Toggle headset mode on and off	Press .
Switch to a handset	Lift the handset.
	Press during a call or after invoking a dial tone.
call	Press Save to preserve the volume level for future calls.

If you use AutoAnswer, see Using AutoAnswer with a Headset or Speakerphone, page 50.

Using a Wireless Headset

Refer to the wireless headset documentation for information about using the wireless headset's remote features. Also, check with your system administrator to be sure your phone is enabled to use a wireless headset with the wireless headset remote hookswitch control feature.

Using Wideband with your Headset

If you use a headset that supports wideband, you might experience improved audio sensitivity if you enable the wideband setting on your phone (this setting is disabled by default). To access the setting, choose \(\subseteq \) > User Preferences > Audio Preferences > Wideband Headset.

If the Wideband Headset setting shows as dimmed, then this setting is not user controllable.

Check with your system administrator to be sure your phone system is configured to use wideband. If the system is not configured for wideband, you may not detect any additional audio sensitivity even when using a wideband headset. To learn more about your headset, refer to the headset documentation or ask your system administrator for assistance.

Using a Speakerphone

Many of the actions you can take to dial a number or answer a call will automatically trigger speakerphone mode, assuming that the handset is in its cradle and () is not lit.

If you want to	Then
Toggle speakerphone mode on or off	Press d.
Switch to a handset	Lift the handset.
Adjust the volume level for a	Press during a call or after invoking a dial tone.
call	Press Save to preserve the volume level for future calls.

Using AutoAnswer with a Headset or Speakerphone

When AutoAnswer is enabled, your phone answers incoming calls automatically after a few rings. Your system administrator configures AutoAnswer to use either the speakerphone or a headset. You might use AutoAnswer if you receive a high volume of incoming calls.

If you	Then
Use AutoAnswer with a headset	Keep headset mode active (in other words, keep (illuminated), even when you are not on a call.
	To keep headset mode active, do the following:
	• Press EndCall to hang up.
	• Press New Call or Dial to place new calls.
	If your phone is set up to use AutoAnswer in headset mode, calls are automatically answered only if () is illuminated.
	Otherwise, calls ring normally and you must manually answer them.
Use AutoAnswer with the speakerphone	Keep the handset in the cradle and headset mode inactive (unlit). Otherwise, calls ring normally and you must manually answer them.

Changing Phone Settings

You can customize your Cisco Unified IP Phone by adjusting the ring tone, background image, and other settings.

Customizing Rings and Message Indicators

You can customize how your phone indicates an incoming call and a new voice mail message. You can also adjust the ringer volume for your phone.

If you want to Then	
Change the ring tone per line	 Choose Subset Viser Preferences > Rings. Choose a phone line or the default ring setting.
	 3. Choose a ring tone to play a sample of it. 4. Press Select and Save to set the ring tone, or press Cancel.
Change the ring pattern per line (flash-only, ring once, beep-only, etc.)	1. Log in to your User Options web pages. (See Accessing Your User Options Web Pages, page 60.)
	2. Access your call ring pattern settings. (See Controlling Line Settings on the Web, page 68.)
	Note Before you can access this setting, your system administrator might need to enable it for you.
Adjust the volume level for the phone ringer	Press while the handset is in the cradle and the headset and speakerphone buttons are off. The new ringer volume is saved automatically.
Change the way that the voice message light on your handset works	1. Log in to your User Options web pages. (See Accessing Your User Options Web Pages, page 60.)
	2. Access your message indicator settings. (See Controlling Line Settings on the Web, page 68.)
	Note Typically, the default system policy is to indicate a new voice message by displaying a steady light on the handset light strip.

Customizing the Phone Screen

You can adjust the characteristics of your phone screen.

If you want to	Then
Change the phone screen contrast	1. Choose S > User Preferences > Contrast.
	2. To make adjustments, press Up, Down or
	3. Press Save, or press Cancel.
	Note If you accidentally save a very light or very dark contrast and cannot see the phone screen display:
	Press and then press 1, 4 on the keypad.
	Next, press to change the contrast until you can see the phone screen display, and then press Save .
Change the background	1. Choose > User Preferences > Background Images.
image	2. Scroll through available images and press Select to choose an image.
	3. Press Preview to see a larger view of the background image.
	4. Press Exit to return to the selection menu.
	5. Press Save to accept the image or press Cancel.
	Note If you do not see a selection of images, then this option has not been enabled on your system.
Change the language on your screen	1. Log in to your User Options web pages. (See Accessing Your User Options Web Pages, page 60.)
	2. Access your user settings. (See Controlling User Settings on the Web, page 67.)
Change the line text label	1. Log in to your User Options web pages. (See Accessing Your User Options Web Pages, page 60.)
	2. Access your line label settings. (See Controlling Line Settings on the Web, page 68.)
Adjust contrast for your Cisco Unified IP Phone Expansion Module	1. Choose S > User Preferences > Contrast.
	2. To make adjustments, press Up, Down or
	3. Press Save, or press Cancel.
	Note The Cisco Unified IP Phone 7942G does not support the Cisco Unified IP Phone Expansion Module.

Using Call Logs and Directories

This section describes how you can use call logs and directories. To access both features, use the Directories button .

Using Call Logs

Your phone maintains logs of your missed, placed, and received calls.

If you want to	Then
View your call logs	Choose > Missed Calls, Placed Calls, or Received Calls. Each stores up to 100 records.
Display details for a	1. Choose > Missed Calls, Placed Calls, or Received Calls.
single call record	2. Highlight a call record.
	3. Press Details. Doing so displays information such as called number, calling number, time of day, and call duration (for placed and received calls only).
Erase all call records in all logs	Press , then press Clear.
Erase all call records in a	1. Choose > Missed Calls, Placed Calls, or Received Calls.
single log	2. Highlight a call record.
	3. Press Clear. (You may need to press the more softkey to display Clear.)
Erase a single call record	1. Choose > Missed Calls, Placed Calls, or Received Calls.
	2. Highlight a call record.
	3. Press Delete.
Dial from a call log	1. Choose > Missed Calls, Placed Calls, or Received Calls.
(while not on another call)	2. Highlight a call record.
	Note If the Details softkey appears, the call is the primary entry of a multiparty call. See the Tips section below.
	3. If you need to edit the displayed number, press EditDial followed by << or >>. To delete the number, press EditDial followed by Delete. (You may need to press the more softkey to display Delete.)
	4 . Go off-hook to place the call.

If you want to	Then
Dial from a call log (while connected to	1. Choose > Missed Calls, Placed Calls, or Received Calls.
	2. Highlight a call record.
another call)	Note If the Details softkey appears, the call is the primary entry of a multiparty call. See the Tips section below.
	3. If you need to edit the displayed number, press EditDial followed by << or >>. To delete the number, press EditDial followed by Delete. (You may need to press the more softkey to display Delete.)
	4. Press Dial.
	5. Choose a menu item to handle the original call:
	Hold—Puts the first call on hold and dials the second.
	• Transfer—Transfers the first party to the second and drops you from the call. (Press Transfer again after dialing to complete the action.)
	• Conference—Creates a conference call with all parties, including you. (Press Confrn or Conference again after dialing to complete the action.)
	EndCall—Disconnects the first call and dials the second.
See if the line in the call log is busy before placing a call to that line	Look for Busy Lamp Field indicators. See Using BLF to Determine a Line State, page 41.
Place a call from a URL	1. Choose > Missed Calls, Placed Calls, or Received Calls.
entry in a call log (SIP phones only)	2. Highlight the URL entry that you want to dial.
phones omy	3. If you need to edit the entry, press EditDial.
	4. The @ icon appears to indicate that you can begin editing characters in the URL entry.
	5. Press Dial.

- (SCCP and SIP phones) Your phone might be set up for international call logging, which is indicated by a "+" symbol on the call logs, redial, or call directory entries. See your system administrator for more information.
- (SCCP phones only) To view the complete call record of a multiparty call (for example, of call that has been forwarded or transferred to you), highlight the call record and press **Details**. The Details record shows two entries for each missed or received multiparty call. The entries are listed in reverse chronological order:

- The first logged entry is the name/number of the last completed call of a multiparty call received on your phone.
- The second logged entry is the name/number of the first completed call of a multiparty call received on your phone.

Directory Dialing

Depending on configuration, your phone can provide corporate and personal directory features:

- Corporate Directory—Corporate contacts that you can access on your phone. Your system administrator sets up and maintains your Corporate Directory.
- Personal Directory—If available, personal contacts and associated speed-dial codes that you can configure and access from your phone and User Options web pages. Personal Directory is comprised of Personal Address Book (PAB) and Fast Dials:
 - PAB is a directory of your personal contacts.
 - Fast Dials allows you to assign codes to PAB entries for quick dialing.

Using Corporate Directory on Your Phone

You can use a corporate directory to place calls to coworkers.

If you want to	Then
Dial from a corporate directory (while not on another call)	1. Choose > Corporate Directory (exact name can vary).
	2. User your keypad to enter a full or partial name and press Search.
	3. To dial, select the listing, or scroll to the listing and go off-hook.
Dial from a corporate	1. Choose > Corporate Directory (exact name can vary).
directory (while on another call)	2. User your keypad to enter a full or partial name and press Search.
another can)	3. Scroll to a listing and press Dial.
	4. Choose a menu item to handle the original call:
	 Hold—Puts the first call on hold and dials the second.
	 Transfer—Transfers the first party to the second and drops you from the call. (Press Transfer again after dialing to complete the action.)
	 Conference—Creates a conference call with all parties, including you. (Press Confrn or Conference again after dialing to complete the action.)
	 EndCall—Disconnects the first call and dials the second.

If you want to	Then
	Look for Busy Lamp Field (BLF) indicators. See Using BLF to Determine a Line State, page 41.

Tip

Use the numbers on your keypad to enter characters on your phone screen. Use the Navigation button on your phone to move between input fields.

Using Personal Directory on Your Phone

The Personal Directory feature set contains your Personal Address Book (PAB) and Fast Dials. This section describes how to set up and use Personal Directory on your phone. Alternately, see Using Personal Directory on the Web, page 61.

If you want to	Then
Access Personal Directory (for PAB and Fast Dial codes)	1. Choose > Personal Directory (exact name can vary).
	2. Enter your Cisco Unified Communications Manager user ID and PIN, then press Submit .
Search for a PAB	1. Access Personal Directory, then choose Personal Address Book.
entry	2. Enter search criteria and press Submit.
	3. You can choose Previous or Next to move through listings.
	4. Highlight the PAB listing that you want and press Select.
Dial from a PAB	1. Search for a listing.
entry	2. Highlight the listing and press Select.
	3. Press Dial. (You may need to press the more softkey to see Dial.)
	4. Enter the participant's phone number.
	5. Highlight the number that you want to dial and press OK.
	6. Press OK again to dial the number.
Delete a PAB entry	1. Search for a listing.
	2. Highlight the listing and press Select.
	3. Press Delete.
	4. Choose OK to confirm the deletion.

If you want to	Then
Edit a PAB entry	1. Search for a listing.
	2. Highlight the listing and press Edit to modify a name or email address.
	3. If necessary, choose Phones to modify a phone number.
	4. Press Update.
Add a new PAB entry	1. Access Personal Directory, then choose Personal Address Book.
	2. Access the Search page by choosing Submit . (You do not need to input search information first.)
	3. Press New.
	4. Use your phone keypad to enter a name and e-mail information.
	5. Choose Phones and use the keypad to enter phone numbers. Be sure to include any necessary access codes such as a 9 or 1.
	6. Choose Submit to add the entry to the database.
Assign a Fast Dial	1. Search for a PAB entry.
code to a PAB entry	2. Highlight the listing and press Select.
	3. Press Fast Dial.
	4. Highlight the number that you want to dial and press Select.
	5. Highlight the Fast Dial code that you want to assign to the number and press Select .
Add a new Fast Dial	1. Choose > Personal Directory > Personal Fast Dials.
code (not using a PAB	2. Press Fast Dial.
entry)	3. Highlight a Fast Dial code that is unassigned and press Select.
	4. Press Assign.
	5. Enter a phone number.
	6. Press Update.
Search for Fast Dial	1. Choose > Personal Directory > Personal Fast Dials.
codes	2. You can choose Previous or Next to move through listings.
	3. Highlight the listing that you want and press Select.
Place a call using a	1. Search for a Fast Dial code.
Fast Dial code	2. Highlight the listing you want and press Select.
	3. Press Dial.
	4. Choose OK to complete the action.

If you want to	Then
Delete a Fast Dial	1. Search for a Fast Dial code.
code	2. Highlight the listing you want and press Select.
	3. Press Remove.
Log out of Personal Directory	1. Choose > Personal Directory (exact name can vary).
	2. Choose Logout.

- Your system administrator can provide you with the user ID and PIN that you need to log in to Personal Directory.
- Personal Directory automatically logs you out after a certain amount of time. This time limit can vary. Ask your system administrator for more information.
- Use the numbers on your keypad to enter characters on your phone screen. Use the Navigation button on your phone to move between input fields.
- Your phone might be set up for international call logging, which is indicated by a "+" symbol on the call logs, redial, or call directory entries. See your system administrator for more information.

Accessing Voice Messages

To access voice messages, use the button.



Your company determines the voice message service that your phone system uses. For the most accurate and detailed information, refer to the documentation that came with your voice message system.

If you want to	Then
Set up and personalize your voice message service	Press and follow the voice instructions.
	If a menu appears on your screen, choose an appropriate menu item.
Check for your new	Look for:
voice messages	• A steady red light on your handset. (This indicator can vary. See Customizing Rings and Message Indicators, page 51.)
	• A flashing message waiting icon and text message on your phone screen.
	Note The red light and message waiting icon display only when you have a voice message on your primary line, even if you receive voice messages on other lines.
	Listen for:
	• A stutter tone from your handset, headset, or speakerphone when you place a call.
	Note The stutter tone is line-specific. You hear it only when using the line with the waiting messages.
Listen to your voice messages or access the voice messages menu	Press .
	Depending on your voice message service, doing so either auto-dials the message service or provides a menu on your screen.
Send a call to a voice message system	Press iDivert. For more information, see Sending a Call to a Voice Message System, page 25.

Using the User Options Web Pages

Your Cisco Unified IP Phone is a network device that can share information with other network devices in your company, including your personal computer. You can use your computer to log in to your Cisco Unified CM User Options web pages, where you can control features, settings, and services for your Cisco Unified IP Phone. For example, you can set up speed-dial buttons from your User Options web pages.

Accessing Your User Options Web Pages

This section describes how to log in and select a phone device.

If you want to	Then do this
Log in to your User Options web pages	1. Obtain a User Options URL, user ID, and default password from your system administrator.
	2. Open a web browser on your computer, enter the URL, and log on.
	3. If prompted to accept security settings, click Yes or Install Certificate.
	The Cisco Unified Communications Manager User Options main web page displays. From this page you can choose User Options to access User Settings, Directory features, a Personal Address Book, and Fast Dials.
	Or, to access phone-specific options, select a device (see below).
logging in	1. After you have logged in to your User Options web pages, choose User Options > Device.
	The Device Configuration page displays.
	2. If you have multiple devices assigned to you, choose the appropriate device (phone model, Extension Mobility profile, or Remote Destination profile) from the Name drop-down menu.
	Note Toolbar buttons located at the top of the Device Configuration page are specific to the selected device type.
Select a configuration option after logging in	1. After you have logged in to your User Options web pages, choose User Options to access User Settings, Directory, Personal Address Book, Fast Dials, and Mobility Settings.
	2. To return to the Device Configuration page from another page, choose User Options > Device.

Configuring Features and Services on the Web

The topics in this section describe how to configure features and services from your User Options web pages after logging in. See Accessing Your User Options Web Pages, page 60.

Using Personal Directory on the Web

The Personal Directory feature set that you can access on your computer consists of:

- A Personal Address Book (PAB)
- Fast Dials
- Cisco Unified Communications Manager Address Book Synchronizer



You can also access PAB and Fast Dials from your phone. See Using Personal Directory on Your Phone, page 56.

Using Your Personal Address Book on the Web

This section describes how to use your PAB from your User Options web pages.

If you want to	Then do this after you log in
Add a new PAB entry	1. Choose User Options > Personal Address Book.
	2. Click Add New.
	3. Enter information for the entry.
	4. Click Save.
Search for a PAB entry	1. Choose User Options > Personal Address Book.
	2. Specify search information and click Find.
Edit a PAB entry	1. Search for a PAB entry.
	2. Click a name or nickname.
	3. Edit the entry as needed and click Save.

If you want to	Then do this after you log in
Delete a PAB entry	1. Search for a PAB entry.
	2. Select one or more entries.
	3. Click Delete Selected.
Assign a line button for PAB	Note Before you can assign a line button for PAB, your system administrator must configure the phone to display services. Contact your system administrator for more information.
	1. Choose User Options > Device.
	2. Click Service URL.
	3. Choose the Personal Address Book service from the Button drop-down list box.
	4. Enter a phone label for the button.
	5. Click Save.
	6. Click Reset and then click Restart to refresh the phone configuration.
	You can now press the line button to access PAB codes.

Configuring Fast Dials on the Web

This section describes how to assign Fast Dials from your User Options web pages.

If you want to	Then do this after you log in
Assign a line button for Fast Dial	Note Before you can assign a line button for Fast Dial, your system administrator must configure the phone to display services. Contact your system administrator for more information.
	1. Choose User Options > Device.
	2. Click Service URL.
	3. Choose the Fast Dial service from the Button drop-down list box.
	4. Enter a phone label for the button.
	5. Click Save.
	6. Click Reset and then click Restart to refresh the phone configuration.
	You can now press the line button to access Fast Dial codes.
Assign a Fast Dial code to a PAB entry	1. Create a PAB entry. See Using Your Personal Address Book on the Web, page 61.
	2. Choose User Options > Fast Dials.
	3. Click Add New.
	4. Use the Search Options area to find the appropriate PAB entry.
	5. Click a phone number in the Search Results area.
	6. Change the Fast Dial code, if desired.
	7. Click Save.
Assign a Fast Dial code to a phone number (without using a PAB entry)	1. Choose User Options > Fast Dials.
	2. Click Add New.
	3. Change the Fast Dial code, if desired.
	4. Enter a phone number.
	5. Click Save.
Search for a Fast Dial entry	1. Choose User Options > Fast Dials.
	2. Specify search information and click Find.

If you want to	Then do this after you log in
Edit a Fast Dial phone number	1. Choose User Options > Fast Dials.
	2. Search for the Fast Dial entry that you want to edit.
	3. Click on a component of the entry.
	4. Change the phone number.
	5. Click Save.
Delete a Fast Dial entry	1. Search for a Fast Dial.
	2. Select one or more entries.
	3. Click Delete Selected.

- You can create up to 500 Fast Dial and PAB entries.
- You can create a new Fast Dial entry without using a PAB entry. Such Fast Dial entries are labeled "raw" in the User Options web pages and do not display a configurable text label.

Using the Address Book Synchronization Tool

You can use the Address Book Synchronization Tool (TABSynch) to synchronize your existing Microsoft Windows Address Book (if applicable) with your PAB. Entries from your Microsoft Windows Address Book will then be accessible on your Cisco Unified IP Phone and User Options web pages. Your system administrator can give you access to TABSynch and provide detailed instructions.

Setting Up Speed Dials on the Web

Depending on configuration, your phone can support several speed-dial features:

- Speed-dial buttons
- Abbreviated Dialing
- Fast Dials



For help using speed-dial features, see Speed Dialing, page 34.

If you want to	Then do this after you log in
Set up speed-dial	1. Choose User Options > Device.
buttons	2. Choose a phone from the Name drop-down menu.
	3. Click Speed Dials.
	4. Enter a number and label for a speed-dial button (programmable button) on your phone.
	5. Click Save.
	Note Your phone uses the ASCII Label field.
Set up Abbreviated	1. Choose User Options > Device.
Dialing	2. Choose a phone from the Name drop-down menu.
	3. Click Speed Dials.
	4. Enter a number and label for an Abbreviated Dialing code.
	5. Click Save.
Set up Fast Dials	See Configuring Fast Dials on the Web, page 63.
	You can also set up Fast Dials on your phone. See Using Personal Directory on Your Phone, page 56.

Setting Up Phone Services on the Web

Phone services can include special phone features, network data, and web-based information (such as stock quotes and movie listings). You must first subscribe to a phone service before accessing it on your phone.

If you want to	Then do this after you log in	
Subscribe to a service	1. Choose User Options > Device.	
	2. Choose a phone from the Name drop-down menu.	
	3. Click Phone Services.	
	4. Click Add New.	
	5. Choose a service from the drop-down list and click Next .	
	6. Change the service label and/or enter additional service information, if available (optional).	
	7. Click Save.	
Search for services	1. Select a device.	
	2. Click Phone Services.	
	3. Click Find.	
Change or end services	1. Search for services.	
	2. Select one or more entries.	
	3. Click Delete Selected.	
Change a service name	1. Search for services.	
	2. Click on the service name.	
	3. Change the information and click Save.	

If you want to	Then do this after you log in
Add a service to an available programmable phone button	 Choose User Options > Device. Choose a phone from the Name drop-down menu. Click Service URL. If you do not see this option, ask your system administrator to configure a service URL button for your phone. Choose a service from the Button Service drop-down list. If you want to rename the service, edit the label fields. Your phone uses the ASCII Label field if the phone does not support double-byte character sets. Click Save. Click Reset to reset your phone (necessary to see the new button label on your phone).
Access a service on your phone	Choose (⊕) > Services.

Controlling User Settings on the Web

User settings include your password, PIN, and language (locale) settings.

If you want to	Then do this after you log in	
Change your password	1. Choose User Options > User Settings.	
	2. In the Browser Password area, enter information.	
	3. Click Save.	
Change your PIN	1. Choose User Options > User Settings.	
	2. In the Phone PIN area, enter information.	
	3. Click Save.	
Change the language (locale) for	1. Choose User Options > User Settings.	
your User Options web pages	2. In the User Locale area, choose an item from the Locale drop-down list.	
	3. Click Save.	
Change the language (locale) for	1. Choose User Options > User Settings.	
your phone screen	2. Choose an item from the User Locale drop-down list.	
	3. Click Save.	

Tip

Your PIN and password allow you to access different features and services. For example, use your PIN to log in to Cisco Extension Mobility or Personal Directory on your phone. Use your password to log in to your User Options web pages and Cisco WebDialer on your personal computer. For more information, ask your system administrator.

Controlling Line Settings on the Web

Line settings affect a specific phone line (directory number) on your phone. Line settings can include call-forwarding, voice message indicators, ring patterns, and line labels.

You can set up other line settings directly on your phone:

- Set up call forwarding for your primary phone line—see Forwarding Calls to Another Number, page 26.
- Change rings, display, and other phone-model specific settings—see Changing Phone Settings, page 51.

If you want to	nen do this after you log in	
Set up call forwarding	1. Choose User Options > Device.	
per line	2. Choose a phone from the Name drop-down menu.	
	3. Click Line Settings.	
	4. If you have more than one directory number (line) assigned to your phone, choose a line from the Line drop-down menu.	
	5. In the Incoming Call Forwarding area, choose call forwarding settings for various conditions.	
	6. Click Save.	
Change the voice	1. Choose User Options > Device.	
message indicator (lamp) setting per line	2. Choose a phone from the Name drop-down menu.	
(famp) setting per fine	3. Click Line Settings.	
	4. If you have more than one directory number (line) assigned to your phone, choose a line from the Line drop-down menu.	
	5. In the Message Waiting Lamp area, choose from various settings.	
	Note Typically, the default message waiting setting prompts your phone to display a steady red light from the handset light strip to indicate a new voice message.	
	6. Click Save.	

If you want to	Then do this after you log in	
Change the audible voice message indicator	1. Choose User Options > Device.	
	2. Choose a phone from the Name drop-down menu.	
setting per line	3. Click Line Settings.	
	4. If you have more than one directory number (line) assigned to your phone, choose a line from the Line drop-down menu.	
	5. In the Audible Message Waiting Indicator area, choose from various settings.	
	Note Typically, the default message waiting setting prompts your phone to display a steady red light from the handset light strip to indicate a new voice message.	
	6. Click Save.	
Change or create a line	1. Choose User Options > Device.	
text label that appears	2. Choose a phone from the Name drop-down menu.	
on your phone screen	3. Click Line Settings.	
	4. If you have more than one directory number (line) assigned to your phone, choose a line from the Line drop-down menu.	
	5. In the Line Text Label area, enter a text label.	
	6. Click Save.	
	Note Your phone uses the ASCII Label field if the phone does not support double-byte character sets.	

Setting Up Phones and Access Lists for Mobile Connect

When using Cisco Mobile Connect, you must add your mobile phone and other phones that you want to use to make and receive calls using the same directory numbers as your desk phone. These phones are called *remote destinations*. You can also define access lists to restrict or allow calls from certain numbers to be sent to your mobile phone.

If you want to	hen do this after you log in	
Create an access list	1. Choose User Options > Mobility Settings > Access Lists.	
	2. Click Add New.	
	3. Enter a name to identify the access list and a description (optional).	
	4. Choose whether the access list will allow or block specified calls.	
	5. Click Save.	
	6. Click Add Member to add phone numbers or filters to the list.	
	7. Select an option from the Filter Mask drop-down list box. You can filter a directory number, calls with restricted caller ID (Not Available), or calls with anonymous caller ID (Private).	
	8. If you select a directory number from the Filter Mask drop-down list box, enter a phone number or filter in the DN Mask field. You can use the following wild cards to define a filter:	
	 X (upper or lower case)—Matches a single digit. For example, 408555123X matches any number between 4085551230 and 4085551239. 	
	 !—Matches any number of digits. For example, 408! matches any number starts with 408. 	
	 #—Used as a single digit for exact match. 	
	9. To add this member to the access list, click Save.	
	10. To save the access list, click Save .	

If you want to	Then do this after you log in
Add a new remote destination	1. Choose User Options > Mobility Settings > Remote Destinations.
	2. Click Add New.
	3. Enter the following information:
	 Name—Enter a name for the mobile (or other) phone.
	 Destination Number—Enter your mobile phone number.
	4. Select your remote destination profile from the drop-down list box. Your remote destination profile contains the settings that apply to remote destinations that you create.
	5. Select the Mobile Phone check box to allow your remote destination to accept a call sent from your desktop phone.
	6. Select the Enable Mobile Connect check box to allow your remote destination to ring simultaneously with your desktop phone.
	7. Choose one of the following options in the Ring Schedule area:
	 All the time—Choose this option if you do not want to impose day and time restrictions on ringing the remote destination.
	 As specified below—Choose this option and select from the following items to set up a ring schedule based on day and time:
	 Select a check box for each day of the week you want to allow calls to ring the remote destination.
	• For each day, select All Day or select the beginning and ending times from the drop-down lists.
	• Select the time zone from the drop-down list box.
	8. Choose one of the following ringing options:
	 Always ring this destination.
	 Ring this destination only if the caller is in the allowed access list that you select.
	 Do not ring this destination if the caller is in the blocked access list that you select.
	Note The ring schedule drop-down list boxes include only the access lists that you have created.
	9. Click Save.

Using Cisco WebDialer

Cisco WebDialer allows you to make calls on your Cisco Unified IP Phone to directory contacts by clicking items in a web browser. Your system administrator must configure this feature for you.

If you want to	Then		
Use WebDialer with your User Options	1. Log in to your User Options web pages. See Accessing Your User Options Web Pages, page 60.		
directory	2. Choose User Options > Directory and search for a coworker.		
	3. Click the number that you want to dial.		
	4. If this is your first time using WebDialer, set up preferences and click Submit. (See the last row in this table for details.)		
	5. If the Make Call page appears, click Dial . (See the last row in this table to learn how to suppress this page in the future, if desired.)		
	The call is now placed on your phone.		
	6. To end a call, click Hangup or hang up from your phone.		
Use WebDialer with another online corporate directory (not your User Options directory)	1. Log in to a WebDialer-enabled corporate directory and search for coworkers.		
	2. Click the number that you want to dial.		
	3. When prompted, enter your user ID and password.		
	4. If this is your first time using WebDialer, set up preferences and click Submit. (See the last row in this table for details.)		
	5. If the Make Call page appears, click Dial . (See the last row in this table to learn how to suppress this page in the future, if desired.)		
	The call is now placed on your phone.		
	6. To end a call, click Hangup or hang up from your phone.		

If you want to	Then	
Log out of WebDialer	Click the logout icon in the Make Call or Hang Up page.	
Set up, view, or change WebDialer preferences	Access the Preferences page.	
	The Preferences page appears the first time that you use WebDialer (after you click the number that you want to dial.)	
	To return to Preferences in the future, click the preferences icon from the Make Call or Hang Up page.	
	The Preferences page contains the following options:	
	• Preferred language—Determines the language used for WebDialer settings and prompts.	
	• Use permanent device—Identifies the Cisco Unified IP Phone and directory number (line) that you will use to place WebDialer calls. If you have one phone with a single line, the appropriate phone and line are automatically selected. Otherwise, choose a phone and/or line. Phones are specified by host name. (To display the host name on your phone, choose > Network Configuration > Host Name.)	
	• Use Extension Mobility—If selected, prompts WebDialer to use the Cisco Unified IP Phone that is associated with your Extension Mobility profile (if available).	
	Do not display call confirmation—If selected, prompts WebDialer to suppress the Make Call page. This page appears by default after you click a phone number in a WebDialer-enabled online directory.	

Understanding Additional Configuration Options

Your system administrator can configure your phone to use specific button and softkey templates along with special services and features, if appropriate. This table provides an overview of some configuration options that you might want to discuss with your phone system administrator based on your calling needs or work environment.



You can locate Phone Guides and other documents listed in this table on the web: http://www.cisco.com/en/US/products/hw/phones/ps379/tsd_products_support_series_home.html

If you	Then	For more information
Need to handle more calls on your phone line	Ask your system administrator to configure your line to support more calls.	Talk to your system administrator or phone support team.
Need more than one phone line	Ask your system administrator to configure one or more additional directory numbers for you.	Talk to your system administrator or phone support team.
Need more speed-dial buttons	First make sure that you are using all of your currently available speed-dial buttons. If you need additional speed-dial buttons, try using Abbreviated Dialing or subscribing to the Fast Dial service. Another option is to attach a Cisco Unified IP Phone Expansion Module to your Cisco Unified IP Phone 7962G.	 See: Speed Dialing, page 34 Cisco Unified IP Phone Expansion Module 7914 Phone Guide. Cisco Unified IP Phone Expansion Module 7915 Phone Guide. Cisco Unified IP Phone Expansion Module 7916 Phone Guide. Note The Cisco Unified IP Phone 7942G does not support the Cisco Unified IP Phone Expansion Module.

If you	Then	For more information
Work with (or work as) an administrative assistant	 Consider using: Cisco Unified Communications Manager Assistant A shared line 	See: • Using a Shared Line, page 38 • Cisco Unified Communications Manager Assistant User Guide
Want to use one extension for several phones	Request a shared line. This allows you to use one extension for your desk phone and lab phone, for example.	See Using a Shared Line, page 38.
Share phones or office space with coworkers	 Call Park to store and retrieve calls without using the transfer feature Call Pickup to answer calls ringing on another phone A shared line to view or join coworkers' calls Cisco Extension Mobility to apply your phone number and user profile to a shared Cisco Unified IP Phone 	Ask your system administrator about these features and see: • Advanced Call Handling, page 34 • Using a Shared Line, page 38 • Using Cisco Extension Mobility, page 45
Answer calls frequently or handle calls on someone's behalf	Ask your system administrator to set up the AutoAnswer feature for your phone.	See Using AutoAnswer with a Headset or Speakerphone, page 50.
Need to make video calls (SCCP phones only)	Consider using Cisco Unified Video Advantage, which enables you to make video calls using your Cisco Unified IP Phone, your personal computer, and an external video camera.	Contact your system administrator for additional assistance and see the Cisco Unified Video Advantage Quick Start Guide and User Guide.
Determine the state of a phone line associated with a speed-dial button, call log, or directory listing on your phone	Ask your administrator to set up the Busy Lamp Field (BLF) feature for your phone.	See Using BLF to Determine a Line State, page 41.
Want to temporarily apply your phone number and settings to a shared Cisco Unified IP Phone	Ask your system administrator about the Cisco Extension Mobility Service.	See Using Cisco Extension Mobility, page 45.

Troubleshooting Your Phone

This section provides troubleshooting information for your Cisco Unified IP Phone.

General Troubleshooting

This section provides information to help you troubleshoot general problems with your phone. For more information, see your system administrator.

Symptom	Explanation	
You cannot hear a dial tone or complete a call	 One or more of the following factors might apply: You must log into the Extension Mobility service. You must enter a client matter code (CMC) or forced authorization code (FAC) after dialing a number. (SCCP phones only) 	
	• Your phone has time-of-day restrictions that prevent you from using some features during certain hours of the day.	
The Settings button is unresponsive	Your system administrator might have disabled on your phone.	
The softkey that you	One or more of the following factors might apply:	
want to use does not appear	You must press more to reveal additional softkeys.	
	• You must change the line state (for example, go off-hook or have a connected call).	
	• Your phone is not configured to support the feature associated with that softkey.	
Barge fails and results in a fast busy tone	You cannot barge an encrypted call if the phone you are using is not configured for encryption. When your barge attempt fails for this reason, your phone plays a fast busy tone.	
You are disconnected from a call that you joined using Barge	You will be disconnected from a call that you have joined using Barge if the call is put on hold, transferred, or turned into a conference call.	
CallBack fails	The other party might have call forwarding enabled.	
The phone shows an error message when you attempt to set up Call Forward All	Your phone may reject your attempt to set up Call Forward All directly on the phone if the target number that you enter would create a Call Forward All loop or would exceed the maximum number of links permitted in a Call Forward All chain (also known as maximum hop count). Ask your system administrator for details.	

Viewing Phone Administration Data

Your system administrator might ask you to access administration data on your phone for troubleshooting purposes.

If you are asked to	Then
Access network configuration data	Choose > Network Configuration and select the network configuration item that you want to view.
Access status data	Choose > Status and select the status item that you want to view.
Access phone model information	Choose > Model Information.
Access phone call and voice quality information	Choose Status > Call Statistics.

Using the Quality Reporting Tool

Your system administrator may temporarily configure your phone with the Quality Reporting Tool (QRT) to troubleshoot performance problems. You can press QRT or Quality Reporting Tool to submit information to your system administrator. Depending on configuration, use the QRT to:

- Immediately report an audio problem on a current call.
- Select a general problem from a list of categories and choose reason codes.

Cisco One-Year Limited Hardware Warranty Terms

There are special terms applicable to your hardware warranty and various services that you can use during the warranty period. Your formal Warranty Statement, including the warranties and license agreements applicable to Cisco software, is available on Cisco.com at the following URL:

http://www.cisco.com/en/US/docs/general/warranty/English/1Y1DEN__.html

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