



# Release Notes for Cisco 2800 Series Integrated Services Routers for Cisco IOS Release 12.4(4)XC

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**First Released:** June 9, 2008  
**Last Revised:** September 24, 2008  
**Cisco IOS Release 12.4(4)XC7**  
**OL-9578-02 Seventh Release**

These release notes describe new features and significant software components for the Cisco 2800 series routers that support the Cisco IOS Release 12.4(4)XC releases. These release notes are updated as needed to describe new memory requirements, new features, new hardware support, software platform deferrals, microcode or modem code changes, related document changes, and any other important changes. Use these release notes with the [Cross-Platform Release Notes for Cisco IOS Release 12.4T](#) and [About Cisco IOS Release Notes](#).

For a list of the software caveats that apply to the Release 12.4(4)XC releases, see the “[Caveats](#)” section on page 11. See also [Caveats for Cisco IOS Release 12.4\(20\)T](#). The online caveats document is updated for every maintenance release.

## Contents

- [System Requirements, page 2](#)
- [New and Changed Information, page 7](#)
- [Limitations and Restrictions, page 11](#)
- [Caveats, page 11](#)
- [Additional References, page 62](#)
- [Notices, page 63](#)



# System Requirements

This section describes the system requirements for Release 12.4(4)XC and includes the following sections:

- [Memory Requirements, page 3](#)
- [Hardware Supported, page 6](#)
- [Determining the Software Version, page 6](#)
- [Upgrading to a New Software Release, page 6](#)
- [Feature Set Tables, page 6](#)

## Memory Requirements

Table 1 describes the memory requirements for the Cisco IOS feature sets supported by Cisco IOS Release 12.4(4)XC on the Cisco 2800 series routers.

**Table 1** Required Memory for the Cisco 2800 Series Routers with Cisco IOS Release 12.4(4)XC

Platform	Image Name	Feature Set	Image	Flash Memory	DRAM
2811 2821 2851	Cisco 2800 ADVANCED ENTERPRISE SERVICES  Cisco 2800 AISK9-AESK9 FEAT SET FACTORY UPG FOR BUNDLES  Cisco 2800 ASK9-AESK9 FEAT SET FACTORY UPG FOR BUNDLES	ADVANCED ENTERPRISE SERVICES  AISK9-AESK9 FEAT SET FACTORY UPG FOR BUNDLES  ASK9-AESK9 FEAT SET FACTORY UPG FOR BUNDLES	c2800nm-adventerprisek9-mz	64	256
	Cisco 2800 INT VOICE/VIDEO, IPIPGW, TDMIP GW AES	INT VOICE/VIDEO, IPIPGW, TDMIP GW AES	c2800nm-adventerprisek9_ivs-mz	64	256
	Cisco 2800 ADVANCED ENTERPRISE SERVICES WITH SNA SWITCHING	ADVANCED ENTERPRISE SERVICES WITH SNA SWITCHING	c2800nm-adventerprisek9_sna-mz	64	256
	Cisco 2800 ADVANCED IP SERVICES  Cisco 2800 SPSK9-AISK9 FEAT SET FACTORY UPG FOR BUNDLES  Cisco 2800 ASK9-AISK9 FEAT SET FACTORY UPG FOR BUNDLES	ADVANCED IP SERVICES  SPSK9-AISK9 FEAT SET FACTORY UPG FOR BUNDLES  ASK9-AISK9 FEAT SET FACTORY UPG FOR BUNDLES	c2800nm-advipservicesk9-mz	64	256
	Cisco 2800 ADVANCED SECURITY	ADVANCED SECURITY	c2800nm-advsecurityk9-mz	64	256
	Cisco 2800 ENTERPRISE BASE W/O CRYPTO	ENTERPRISE BASE W/O CRYPTO	c2800nm-entbase-mz	64	256
	Cisco 2800 ENTERPRISE BASE	ENTERPRISE BASE	c2800nm-entbasek9-mz	64	256
	Cisco 2800 ENTERPRISE SERVICES W/O CRYPTO	ENTERPRISE SERVICES W/O CRYPTO	c2800nm-entservices-mz	64	256
	Cisco 2800 ENTERPRISE SERVICES  Cisco 2800 SPSK9-ESK9 FEAT SET FACTORY UPG FOR BUNDLES	ENTERPRISE SERVICES  SPSK9-ESK9 FEAT SET FACTORY UPG FOR BUNDLES	c2800nm-entservicesk9-mz	64	256

**Table 1**      ***Required Memory for the Cisco 2800 Series Routers with Cisco IOS Release 12.4(4)XC (continued)***

<b>Platform</b>	<b>Image Name</b>	<b>Feature Set</b>	<b>Image</b>	<b>Flash Memory</b>	<b>DRAM</b>
	Cisco 2800 IP BASE W/O CRYPTO	IP BASE W/O CRYPTO	c2800nm-ipbase-mz	64	256
	Cisco 2800 IP BASE	IP BASE	c2800nm-ipbasek9-mz	64	256
	Cisco 2800 IP VOICE W/O CRYPTO	IP VOICE W/O CRYPTO	c2800nm-ipvoice-mz	64	256
	Cisco 2800 INT VOICE/VIDEO, IPIP GW, TDMIP GW	INT VOICE/VIDEO, IPIP GW, TDMIP GW	c2800nm-ipvoice_ivs-mz	64	256
	Cisco 2800 IP VOICE	IP VOICE	c2800nm-ipvoicek9-mz	64	256
	Cisco 2800 SP SERVICES	SP SERVICES	c2800nm-spservicesk9-mz	64	256

**Table 1**      **Required Memory for the Cisco 2800 Series Routers with Cisco IOS Release 12.4(4)XC (continued)**

Platform	Image Name	Feature Set	Image	Flash Memory	DRAM
2801	Cisco 2801 IOS AISK9-AESK9 FEAT SET FACTORY UPG FOR BUNDLES	IOS AISK9-AESK9 FEAT SET FACTORY UPG FOR BUNDLES	c2801-adventerprisek9-mz	64	192
	Cisco 2801 IOS ASK9-AESK9 FEAT SET FACTORY UPG FOR BUNDLES	IOS ASK9-AESK9 FEAT SET FACTORY UPG FOR BUNDLES			
	Cisco 2801 IOS INT VOICE/VIDEO, IPIPGW, TDMIP GW AES	IOS INT VOICE/VIDEO, IPIPGW, TDMIP GW AES	c2801-adventerprisek9_ivs-mz	64	256
	Cisco 2801 IOS ASK9-AISK9 FEAT SET FACTORY UPG FOR BUNDLES	IOS SPSK9-AISK9 FEAT SET FACTORY UPG FOR BUNDLES	c2801-advipservicesk9-mz	64	92
	Cisco 2801 IOS SPSK9-AISK9 FEAT SET FACTORY UPG FOR BUNDLES	IOS ASK9-AISK9 FEAT SET FACTORY UPG FOR BUNDLES			
	Cisco 2801 IOS ENTERPRISE BASE W/O CRYPTO	IOS ENTERPRISE BASE W/O CRYPTO	c2801-entbase-mz	64	128
	Cisco 2801 IOS ENTERPRISE BASE	IOS ENTERPRISE BASE	c2801-entbasek9-mz	64	128
	Cisco 2801 IOS ENTERPRISE SERVICES W/O CRYPTO	IOS ENTERPRISE SERVICES W/O CRYPTO	c2801-entservices-mz	64	192
	Cisco 2801 IOS ENTERPRISE SERVICES	Cisco 2801 IOS ENTERPRISE SERVICES	c2801-entservicesk9-mz	64	192
	Cisco 2801 IOS SPSK9-ESK9 FEAT SET FACTORY UPG FOR BUNDLES	IOS SPSK9-ESK9 FEAT SET FACTORY UPG FOR BUNDLES			
	Cisco 2801 IOS IP BASE W/O CRYPTO	IOS IP BASE W/O CRYPTO	c2801-ipbase-mz	64	128
	Cisco 2801 IOS IP BASE	IOS IP BASE	c2801-ipbasek9-mz	64	128
	Cisco 2801 IOS IP VOICE W/O CRYPTO	IOS IP VOICE W/O CRYPTO	c2801-ipvoice-mz	64	192

**Table 1**      **Required Memory for the Cisco 2800 Series Routers with Cisco IOS Release 12.4(4)XC (continued)**

Platform	Image Name	Feature Set	Image	Flash Memory	DRAM
	Cisco 2801 IOS INT VOICE/VIDEO, IPIP GW, TDMIP GW	IOS INT VOICE/VIDEO, IPIP GW, TDMIP GW	c2801-ipvoice_ivs-mz	64	256
	Cisco 2801 IOS IP VOICE	IOS IP VOICE	c2801-ipvoicek9-mz	64	192
	Cisco 2801 IOS SP SERVICES	IOS SP SERVICES	c2801-spservicesk9-mz	64	192
	Cisco 2801 IOS SPSK9-AESK9 FEAT SET FACTORY UPG FOR BUNDLES	IOS SPSK9-AESK9 FEAT SET FACTORY UPG FOR BUNDLES			

## Hardware Supported

Cisco IOS Release 12.4(4)XC supports the following Cisco 2800 series routers:

- Cisco 2801
- Cisco 2811
- Cisco 2821
- Cisco 2851

For descriptions of existing hardware features and supported modules, see the hardware installation guides, configuration and command reference guides, and additional documents specific to the Cisco 2800 series routers, which are available at:

[http://www.cisco.com/en/US/products/ps5854/tsd\\_products\\_support\\_series\\_home.html](http://www.cisco.com/en/US/products/ps5854/tsd_products_support_series_home.html)

## Determining the Software Version

To determine the version of Cisco IOS software currently running on your Cisco 2800 series router, see *About Cisco IOS Release Notes* located at:

[http://www.cisco.com/en/US/docs/ios/12\\_4/12\\_4x/12\\_4xy15/ReleaseNote.html](http://www.cisco.com/en/US/docs/ios/12_4/12_4x/12_4xy15/ReleaseNote.html).

## Upgrading to a New Software Release

For general information about upgrading to a new software release, see *About Cisco IOS Release Notes* located at:

[http://www.cisco.com/en/US/docs/ios/12\\_4/12\\_4x/12\\_4xy15/ReleaseNote.html](http://www.cisco.com/en/US/docs/ios/12_4/12_4x/12_4xy15/ReleaseNote.html).

## Feature Set Tables

For information about feature set tables, see *About Cisco IOS Release Notes* located at:

[http://www.cisco.com/en/US/docs/ios/12\\_4/12\\_4x/12\\_4xy15/ReleaseNote.html](http://www.cisco.com/en/US/docs/ios/12_4/12_4x/12_4xy15/ReleaseNote.html).

# New and Changed Information

This section contains the following information:

- [Updated Naming Conventions](#), page 7
- [New Software Features in Release 12.4\(4\)XC4](#), page 7
- [New Software Features in Release 12.4\(4\)XC1](#), page 7
- [New Hardware Features in Release 12.4\(4\)XC](#), page 7
- [New Software Features in Release 12.4\(4\)XC](#), page 8

## Updated Naming Conventions

The following product names have been changed in Cisco IOS Release 12.4(4)XC:

- Cisco Unified Communications Manager Express (Cisco CME) is now Cisco Unified Communications Manager Express. This change is effective in Cisco Unified CME version 4.0 and later releases.
- Cisco Survivable Remote Site Telephony (Cisco SRST) is now Cisco Unified SRST. This change is effective with Cisco Unified SRST version 4.0 and later releases.

## New Software Features in Release 12.4(4)XC4

### Cisco Unified CME 4.0(4) Extension Assigner

The Cisco Unified Communications Manager Express (CME) feature enables installation technicians to assign extension numbers to Cisco Unified CME phones without accessing the server. For more information, see the following URL:

[http://www.cisco.com/en/US/docs/voice\\_ip\\_comm/cucme/admin/configuration/guide/cmexasgn.html](http://www.cisco.com/en/US/docs/voice_ip_comm/cucme/admin/configuration/guide/cmexasgn.html)

## New Software Features in Release 12.4(4)XC1

New features are available for Cisco Unified Communications Manager Express (Cisco Unified CME), including the following:

- FXO trunk enhancements—Supports shared lines, transfer recall for transferred and forwarded calls, status monitoring of FXO ports, and line button optimization for call transfer.
- Automatic line selection enhancement—Enables automatic line selection for incoming calls on the line associated with a specified button.
- Night service ring enhancement—Overrides silent ringing during active night service periods.

For more information, see the [“Cisco Unified Communications Manager Express 4.0”](#) section on page 8.

## New Hardware Features in Release 12.4(4)XC

There are no new hardware features in Cisco IOS Release 12.4(4)XC.

## New Software Features in Release 12.4(4)XC

The following sections describe the new software features supported by the Cisco 2800 series routers for Cisco IOS Release 12.4(4)XC:

- [Call Detail Records \(CDR\) Feature Correlation ID for Supplementary Features, page 8](#)
- [Cisco Unified Communications Manager Express 4.0, page 8](#)
- [Cisco Unified Survivable Remote Site Telephony 4.0, page 9](#)
- [Customizable PSTN Tones and H.323 Call-Disconnect Cause Codes, page 9](#)
- [H.323 VoIP Call Preservation Enhancements for WAN Link Failures, page 10](#)
- [Integrated Services, page 10](#)
- [SCCP Analog \(FXS\) Ports, page 10](#)
- [SCCP PLAR with DTMF Out Pulse Digits for FXS Analog Phones, page 10](#)
- [Secure Communication Between IP-STE Endpoint and Lineside STE Endpoint, page 11](#)
- [Secure Communication Between IP-STE Endpoint and Trunkside STE Endpoint, page 11](#)
- [Video Support for SCCP-Based Endpoints, page 11](#)

### Call Detail Records (CDR) Feature Correlation ID for Supplementary Features

This feature captures additional information in CDRs for voice calls that are transferred or forwarded on phones controlled by Cisco Unified Communications Manager Express (CME) or Cisco Unified Survivable Remote Site Telephony (SRST). It includes a unique correlation ID that identifies a given call feature across all legs in a call. CDR information can be output in RADIUS VSAs or system log (syslog) messages. For more information about this feature, go to the following URL:

[http://www.cisco.com/en/US/docs/ios/12\\_4t/12\\_4t11/ht\\_fcid.html](http://www.cisco.com/en/US/docs/ios/12_4t/12_4t11/ht_fcid.html)

### Cisco Unified Communications Manager Express 4.0

Cisco Unified Communications Manager Express 4.0 (Cisco Unified CME 4.0) delivers a number of next-generation telephony features that expand system capabilities and provide key productivity enhancements.

With this release, the product name has been changed to Cisco Unified Communications Manager Express to underscore its position as a member of the Cisco Unified Communications family of products.

New features and expanded capabilities are introduced in the following areas:

- Basic Automatic Call Distribution (B-ACD) and Auto-Attendant (AA) Service
- Direct Inward Dial Digit Translation Service, which provides a number transformation service for DID calls
- Call forwarding, call park, call transfer, and conferencing
- Ephone hunt groups, including dynamic membership and agent status control
- Ephone templates and ephone-dn templates to apply features quickly to phones and lines
- Phone support for remote teleworkers, Cisco IP Communicator, and the newest Cisco Unified IP phones:
  - Cisco Unified IP Phone 7911G

- Cisco Unified IP Phone 7941G
- Cisco Unified IP Phone 7941G-GE
- Cisco Unified IP Phone 7961G
- Cisco Unified IP Phone 7961G-GE
- IP phone authentication for secure Skinny Client Control Protocol (SCCP) signaling between Cisco Unified CME and IP phones
- Fax passthrough mode using Cisco VG 224 voice gateways, Analog Telephone Adaptors (ATA), and SCCP
- Video support
- QSIG integration with TDM PBXs
- Feature access codes and feature control
- Redundant Cisco Unified CME router

For details about new features, see the “Feature History” chapter of the Cisco Unified Communications Manager Express System Administrator Guide at:

[http://www.cisco.com/en/US/docs/voice\\_ip\\_comm/cucme/admin/configuration/guide/cmeroad.html](http://www.cisco.com/en/US/docs/voice_ip_comm/cucme/admin/configuration/guide/cmeroad.html)

For links to all Cisco Unified Communications Manager Express documents, see:

[http://www.cisco.com/en/US/products/sw/voicesw/ps4625/tsd\\_products\\_support\\_series\\_home.html](http://www.cisco.com/en/US/products/sw/voicesw/ps4625/tsd_products_support_series_home.html)

## Cisco Unified Survivable Remote Site Telephony 4.0

Cisco Unified Survivable Remote Site Telephony (SRST) 4.0 adds these features to the Cisco 2800 series routers in Cisco IOS Release 12.4(4)XC:

- Support for the following Cisco Unified IP Phone models:
  - Cisco Unified IP Phone 7911G
  - Cisco Unified IP Phone 7941G
  - Cisco Unified IP Phone 7941G-GE
  - Cisco Unified IP Phone 7961G
  - Cisco Unified IP Phone 7961G-GE
- Cisco IP Communicator support
- Fax pass-through for ATA and Cisco VG 224 and Cisco VG 248 using SCCP mode.
- H.323 VoIP call preservation enhancements for WAN link failures
- Video support

For more information about these features, go to

[http://www.cisco.com/en/US/products/sw/voicesw/ps2169/products\\_configuration\\_guide\\_chapter09186a00806672f6.html#wp1048642](http://www.cisco.com/en/US/products/sw/voicesw/ps2169/products_configuration_guide_chapter09186a00806672f6.html#wp1048642).

## Customizable PSTN Tones and H.323 Call-Disconnect Cause Codes

The Customizable PSTN Tones and H.323 Call-Disconnect Cause Codes feature enables you to customize PSTN tones and H.323 call-disconnect cause codes for certain disconnect scenarios. Specifically, you can customize the following:

- PSTN tones that are applicable to foreign-exchange-station (FXS), PRI, and BRI calls and IP phones

- Q.850 call-disconnect cause codes for H.323 gateways

In addition, you can specify the mechanism for detecting media inactivity (silence) on a voice call: Real-Time Transport Protocol (RTP), RTP Control Protocol (RTCP), or both.

## H.323 VoIP Call Preservation Enhancements for WAN Link Failures

Changes made to the Cisco IOS H.323 voice gateway and Cisco Unified Communications Manager Express in support of this feature make it possible to preserve calls when the WAN link flaps, resulting in a temporary TCP connection loss between the Cisco Unified Communications Manager Express and an IP phone or Cisco IOS H.323 voice gateway located in a remote site or branch office location.

## Integrated Services

This feature allows data PRI services (dial-in, dial-on-demand routing [DDR], and DDR backup) to occur on top of voice-enabled PRI interfaces, and adds multilevel precedence and preemption (MLPP) capability for DDR calls over the active voice call when no idle channel is available during the DDR call setup.

- An ISDN interface can now be configured accept multiple call types to allow integrated data and voice services.
- Multilevel precedence and preemption (MLPP) is the placement of priority calls through the network. Precedence designates the priority level that is associated with a call. Preemption designates the process of terminating lower-priority calls so that a call of higher precedence can be extended.

For more information about the Integrated Services feature, see the *Integrating Data and Voice Services for ISDN PRI Interfaces on Multiservice Access Routers* document at:

[http://www.cisco.com/en/US/docs/ios/12\\_4t/12\\_4t11/intserv.html](http://www.cisco.com/en/US/docs/ios/12_4t/12_4t11/intserv.html)

## SCCP Analog (FXS) Ports

This feature provides Cisco IOS software support to enable Skinny Client Control Protocol (SCCP) supplementary features on analog FXS ports on a Cisco voice gateway under the control of Cisco Unified Communications Manager or a Cisco Unified Communications Manager Express (Cisco Unified CME) system. This feature is now supported on Cisco 280x and Cisco 380x platforms in Cisco IOS Release 12.4(4)XC. For more information, see:

[http://www.cisco.com/en/US/docs/ios/12\\_4t/12\\_4t2/ht1vg224.html](http://www.cisco.com/en/US/docs/ios/12_4t/12_4t2/ht1vg224.html)

## SCCP PLAR with DTMF Out Pulse Digits for FXS Analog Phones

This feature provides private line automatic ring-down (PLAR) support for Skinny Client Control Protocol (SCCP) analog ports on a Cisco ISR voice gateway under the control of Cisco Unified Communications Manager or a Cisco Unified Communications Manager Express (Cisco Unified CME) system.

## Secure Communication Between IP-STE Endpoint and Lineside STE Endpoint

This feature allows Secure Telephone Equipment (STE) and Secure Telephone Units (STUs) to encrypt voice and data streams with government proprietary algorithms (Type-1 encryption). To provide support for the legacy STEs & STUs and newer IP-STE, Cisco gateways support voice and data in secure and non-secure modes within the IP network and can pass calls to/from government TDM voice networks. This feature allows analog STUs (Secure Telephony Units) and analog/BRI STEs (Secure Terminal equipment) to communicate in secure mode using v.150.1 modem relay. This feature provides the support for modem relay on gateways to the line side.

## Secure Communication Between IP-STE Endpoint and Trunkside STE Endpoint

This feature enables encrypted and decrypted calls from an IP-STE controlled by Cisco Unified Communications Manager through a voice gateway to an STE in the Defense Switch Network (DSN). This project implements a subset of the v.150.1 modem relay standard.

## Video Support for SCCP-Based Endpoints

The Video Support for SCCP-Based Endpoints feature adds video support for Cisco Unified Communications Manager Express to maintain close feature parity with Cisco Unified Communications Manager. This feature allows you to pass a video stream with a voice call between video-capable SCCP endpoints and between SCCP and H.323 endpoints. Through the Cisco Unified CME router, the video-capable endpoints can communicate with each other locally to a remote H.323 endpoint through a gateway or through an H.323 network.

For information about video support in Cisco Unified Communications Manager Express, see the “Video Support for SCCP-Based Endpoints” chapter in the *Cisco Communications Manager Express System Administrator Guide* at:  
[http://www.cisco.com/en/US/docs/voice\\_ip\\_comm/cucme/admin/configuration/guide/cmevideo.html](http://www.cisco.com/en/US/docs/voice_ip_comm/cucme/admin/configuration/guide/cmevideo.html)

## Limitations and Restrictions

There are no known limitations or restrictions.

## Caveats

For general information on caveats and the bug toolkit, see *About Cisco IOS Release Notes* located at:  
[http://www.cisco.com/en/US/docs/ios/12\\_4/12\\_4x/12\\_4xy15/ReleaseNote.html](http://www.cisco.com/en/US/docs/ios/12_4/12_4x/12_4xy15/ReleaseNote.html).

This section contains the following caveat information:

- [Resolved Caveats - Cisco IOS Release 12.4\(4\)XC7, page 12](#)
- [Open Caveats - Cisco IOS Release 12.4\(4\)XC7, page 18](#)
- [Resolved Caveats - Cisco IOS Release 12.4\(4\)XC7, page 12](#)
- [Open Caveats - Cisco IOS Release 12.4\(4\)XC6, page 25](#)
- [Resolved Caveats - Cisco IOS Release 12.4\(4\)XC5, page 25](#)
- [Open Caveats - Cisco IOS Release 12.4\(4\)XC5, page 33](#)
- [Resolved Caveats - Cisco IOS Release 12.4\(4\)XC4, page 33](#)

- [Open Caveats - Cisco IOS Release 12.4\(4\)XC4, page 40](#) -
- [Resolved Caveats - Cisco IOS Release 12.4\(4\)XC3, page 40](#)
- [Open Caveats - Cisco IOS Release 12.4\(4\)XC3, page 44](#)-
- [Resolved Caveats - Cisco IOS Release 12.4\(4\)XC2, page 44](#)
- [Open Caveats - Cisco IOS Release 12.4\(4\)XC2, page 48](#)
- [Resolved Caveats - Cisco IOS Release 12.4\(4\)XC1, page 48](#)
- [Open Caveats - Cisco IOS Release 12.4\(4\)XC1, page 52](#)
- [Resolved Caveats - Release 12.4\(4\)XC, page 53](#)
- [Open Caveats - Release 12.4\(4\)XC, page 56](#)
- [Caveat Updates and Special Notices, page 61](#)

## Resolved Caveats - Cisco IOS Release 12.4(4)XC7

CSCec12299

Devices running Cisco IOS versions 12.0S, 12.2, 12.3 or 12.4 and configured for Multiprotocol Label Switching (MPLS) Virtual Private Networks (VPNs) or VPN Routing and Forwarding Lite (VRF Lite) and using Border Gateway Protocol (BGP) between Customer Edge (CE) and Provider Edge (PE) devices may permit information to propagate between VPNs.

Workarounds are available to help mitigate this vulnerability.

This issue is triggered by a logic error when processing extended communities on the PE device.

This issue cannot be deterministically exploited by an attacker.

Cisco has released free software updates that address these vulnerabilities. Workarounds that mitigate these vulnerabilities are available.

This advisory is posted at <http://www.cisco.com/warp/public/707/cisco-sa-20080924-vpn.shtml>.

CSCsd81407

Multiple voice-related vulnerabilities are identified in Cisco IOS software, one of which is also shared with Cisco Unified Communications Manager. These vulnerabilities pertain to the following protocols or features:

- Session Initiation Protocol (SIP)
- Media Gateway Control Protocol (MGCP)
- Signaling protocols H.323, H.254
- Real-time Transport Protocol (RTP)
- Facsimile reception

Cisco has made free software available to address these vulnerabilities for affected customers. Fixed Cisco IOS software listed in the Software Versions and Fixes section contains fixes for all vulnerabilities mentioned in this advisory.

There are no workarounds available to mitigate the effects of any of the vulnerabilities apart from disabling the protocol or feature itself.

This advisory is posted at  
<http://www.cisco.com/warp/public/707/cisco-sa-20070808-IOS-voice.shtml>

CSCsi80749

Multiple voice-related vulnerabilities are identified in Cisco IOS software, one of which is also shared with Cisco Unified Communications Manager. These vulnerabilities pertain to the following protocols or features:

- Session Initiation Protocol (SIP)
- Media Gateway Control Protocol (MGCP)
- Signaling protocols H.323, H.254
- Real-time Transport Protocol (RTP)
- Facsimile reception

Cisco has made free software available to address these vulnerabilities for affected customers. Fixed Cisco IOS software listed in the Software Versions and Fixes section contains fixes for all vulnerabilities mentioned in this advisory.

There are no workarounds available to mitigate the effects of any of the vulnerabilities apart from disabling the protocol or feature itself.

This advisory is posted at  
<http://www.cisco.com/warp/public/707/cisco-sa-20070808-IOS-voice.shtml>

CSCsg70474

Multiple voice-related vulnerabilities are identified in Cisco IOS software, one of which is also shared with Cisco Unified Communications Manager. These vulnerabilities pertain to the following protocols or features:

- Session Initiation Protocol (SIP)
- Media Gateway Control Protocol (MGCP)
- Signaling protocols H.323, H.254
- Real-time Transport Protocol (RTP)
- Facsimile reception

Cisco has made free software available to address these vulnerabilities for affected customers. Fixed Cisco IOS software listed in the Software Versions and Fixes section contains fixes for all vulnerabilities mentioned in this advisory.

There are no workarounds available to mitigate the effects of any of the vulnerabilities apart from disabling the protocol or feature itself.

This advisory is posted at  
<http://www.cisco.com/warp/public/707/cisco-sa-20070808-IOS-voice.shtml>

CSCsi60004

Multiple voice-related vulnerabilities are identified in Cisco IOS software, one of which is also shared with Cisco Unified Communications Manager. These vulnerabilities pertain to the following protocols or features:

- Session Initiation Protocol (SIP)

- Media Gateway Control Protocol (MGCP)
- Signaling protocols H.323, H.254
- Real-time Transport Protocol (RTP)
- Facsimile reception

Cisco has made free software available to address these vulnerabilities for affected customers. Fixed Cisco IOS software listed in the Software Versions and Fixes section contains fixes for all vulnerabilities mentioned in this advisory.

There are no workarounds available to mitigate the effects of any of the vulnerabilities apart from disabling the protocol or feature itself.

This advisory is posted at

<http://www.cisco.com/warp/public/707/cisco-sa-20070808-IOS-voice.shtml>

- CSCse56501

A device running Cisco IOS software that has Internet Protocol version 6 (IPv6) enabled may be subject to a denial of service (DoS) attack. For the device to be affected by this vulnerability the device also has to have certain Internet Protocol version 4 (IPv4) User Datagram Protocol (UDP) services enabled. To exploit this vulnerability an offending IPv6 packet must be targeted to the device. Packets that are routed throughout the router can not trigger this vulnerability. Successful exploitation will prevent the interface from receiving any additional traffic. The only exception is Resource Reservation Protocol (RSVP) service, which if exploited, will cause the device to crash. Only the interface on which the vulnerability was exploited will be affected.

Cisco is providing fixed software to address this issue. There are workarounds available to mitigate the effects of the vulnerability.

This advisory is posted at

<http://www.cisco.com/warp/public/707/cisco-sa-20080326-IPv4IPv6.shtml>.

CSCsg96319 reverse ssh eliminated telnet authentication on VTY

**Symptom** When a reverse SSH session is established with valid authentication credentials, anyone can obtain unprivileged Telnet access to a system without being authenticated. This situation affects only reverse SSH sessions when a connection is made with the

**ssh -l *userid* :*number* *ip-address*** command.

**Conditions** This symptom is observed only when the Reverse SSH Enhancement is configured. This enhancement is documented at the following URL:

[http://www.cisco.com/en/US/docs/ios/12\\_3t/12\\_3t11/feature/guide/gt\\_rssh.html](http://www.cisco.com/en/US/docs/ios/12_3t/12_3t11/feature/guide/gt_rssh.html)

**Workaround** Configure reverse SSH by entering the **ip ssh port *portnum* rotary group** command. This configuration is explained at the following URL:

[http://www.cisco.com/en/US/tech/tk583/tk617/technologies\\_q\\_and\\_a\\_item09186a0080267e0f.shtml#newq1](http://www.cisco.com/en/US/tech/tk583/tk617/technologies_q_and_a_item09186a0080267e0f.shtml#newq1)

CSCsg40567 Memory leak found with malformed tls/ssl packets in http core process

**Symptom** Malformed SSL packets may cause a router to leak multiple memory blocks.

**Conditions** This symptom is observed on a Cisco router that has the **ip http secure server** command enabled.

**Workaround** Disable the **ip http secure server** command.

CSCsg03449 Etherswitch module VLAN Trunking Protocol Vulnerabilities

**Symptom**

- VTP Version field DoS
- Integer Wrap in VTP revision
- Buffer Overflow in VTP VLAN name

**Conditions** The packets must be received on a trunk enabled port.

**Further Information:** On the 13th September 2006, Phenoelit Group posted an advisory containing three vulnerabilities:

- VTP Version field DoS
- Integer Wrap in VTP revision
- Buffer Overflow in VTP VLAN name

These vulnerabilities are addressed by Cisco IDs:

- [CSCsd52629/CSCsd34759](#) -- VTP version field DoS
- [CSCse40078/ CSCse47765](#) -- Integer Wrap in VTP revision
- [CSCsd34855/CSCei54611](#) -- Buffer Overflow in VTP VLAN name
- [CSCsg03449](#) -- Etherswitch module VLAN Trunking Protocol Vulnerabilities. Cisco's statement and further information are available on the Cisco public website at: <http://www.cisco.com/warp/public/707/cisco-sr-20060913-vtp.shtml>

CSCsj44099 Router crashes if DSPFARM profile description is 128 characters long.

**Symptom** A cisco c3800 router can experience a memory corruption resulting in a crash if the description field under the “dspfarm profile” configuration matches the maximum of 128 characters.

**Conditions** During configuration of the dspfarm profile through the CLI, a description that is 128 characters will cause a memory copy problem. If the user tries to display the results of the configuration using “show dspfarm profile”, the router will crash trying to display the output.

**Workaround** To prevent this problem configure the dspfarm profile description with 127 characters or less.

CSCse05736 A router running RCP can be reloaded with a specific packet

**Symptom** A router that is running RCP can be reloaded by a specific packet.

**Conditions** This symptom is seen under the following conditions

- The router must have RCP enabled.
- The packet must come from the source address of the designated system configured to send RCP packets to the router.
- The packet must have a specific data content.

**Workaround** Put access lists on the edge of your network blocking RCP packets to prevent spoofed RSH packets. Use another protocol such as SCP. Use VTY Acts.

CSCec12299 Corruption of ext communities when receiving over ipv4 EBGp session

**Symptom** EIGRP-specific Extended Community 0x8800 is corrupted and shown as 0x0:0:0.

**Conditions** This symptom is observed when EIGRP-specific Extended Community 0x8800 is received via an IPv4 EBGp session on a CE router. This occurs typically in the following inter-autonomous system scenario:

**ASBR/PE-1 <----> VRF-to-VRF <----> ASBR/PE-2**

**Workaround** Use a configuration such as the following to remove extended communities from the CE router:

```
router bgp 1
 address-family ipv4 vrf one
  neighbor 1.0.0.1 remote-as 100
  neighbor 1.0.0.1 activate
  neighbor 1.0.0.1 route-map FILTER in
 exit-address-family
!
ip extcommunity-list 100 permit _RT.*_
!
!
route-map FILTER permit 10
 set extcomm-list 100 delete
!
```

CSCse24889 Malformed SSH version 2 packets may cause processor memory depletion

**Symptom** Malformed SSH version 2 packets may cause a memory leak, causing the platform to operate under a degraded condition. Under rare circumstances, the platform may reload to recover itself.

**Conditions** This symptom is observed on a Cisco platform that is configured for SSH version 2 after it has received malformed SSHv2 packets.

**Workaround** As an interim solution until the affected platform can be upgraded to a Cisco IOS software image that contains the fix for caveat CSCse24889, configure SSH version 1 from the global configuration mode, as in the following example:

```
config t

ip ssh version 1
end
```

**Alternate Workaround:** Permit only known trusted hosts and/or networks to connect to the router by creating a vty access list, as in the following example:

#### Workaround

```
10.1.1.0/24 is a trusted network that
is permitted access to the router, all
other access is denied
```

```
access-list 99 permit 10.1.1.0 0.0.0.255
access-list 99 deny any
```

```
line vty 0 4
access-class 99 in
end
```

**Further Problem Description:** For information about configuring vty access lists, see the [Controlling Access to a Virtual Terminal Line](#) document.

For information about SSH, see the Configuring Secure Shell on Routers and Switches Running Cisco IOS document: <http://www.cisco.com/warp/public/707/ssh.shtml>

CSCsc40493 Lengthy PADR frame could crash PPPoE BRAS

**Symptom** A PPPoE aggregation server (BRAS) may reset when receiving a malformed PPPoE message.

**Conditions** A malformed PPPoE message must be received on an aggregation interface.

**Workaround** There is no workaround.

CSCsh53643 mbar/isync compiler automation (No RNE available)

CSCsh77241 Reverting the compiler back to c2.95.3-p11b (No RNE available)

- CSCsi01470

A vulnerability in the Cisco implementation of Multicast Virtual Private Network (MVPN) is subject to exploitation that can allow a malicious user to create extra multicast states on the core routers or receive multicast traffic from other Multiprotocol Label Switching (MPLS) based Virtual Private Networks (VPN) by sending specially crafted messages.

Cisco has released free software updates that address this vulnerability. Workarounds that mitigate this vulnerability are available.

This advisory is posted at <http://www.cisco.com/warp/public/707/cisco-sa-20080326-mvpn.shtml>.

## Open Caveats - Cisco IOS Release 12.4(4)XC7

There are no open caveats in this release.

## Resolved Caveats - Cisco IOS Release 12.4(4)XC6

### CSCsf30058

Multiple voice-related vulnerabilities are identified in Cisco IOS software, one of which is also shared with Cisco Unified Communications Manager. These vulnerabilities pertain to the following protocols or features:

- Session Initiation Protocol (SIP)
- Media Gateway Control Protocol (MGCP)
- Signaling protocols H.323, H.254
- Real-time Transport Protocol (RTP)
- Facsimile reception

Cisco has made free software available to address these vulnerabilities for affected customers. Fixed Cisco IOS software listed in the Software Versions and Fixes section contains fixes for all vulnerabilities mentioned in this advisory.

There are no workarounds available to mitigate the effects of any of the vulnerabilities apart from disabling the protocol or feature itself.

This advisory is posted at

<http://www.cisco.com/warp/public/707/cisco-sa-20070808-IOS-voice.shtml>

### CSCsb40304

Cisco IOS device may crash while processing malformed Secure Sockets Layer (SSL) packets. In order to trigger these vulnerabilities, a malicious client must send malformed packets during the SSL protocol exchange with the vulnerable device.

Successful repeated exploitation of any of these vulnerabilities may lead to a sustained Denial-of-Service (DoS); however, vulnerabilities are not known to compromise either the confidentiality or integrity of the data or the device. These vulnerabilities are not believed to allow an attacker will not be able to decrypt any previously encrypted information.

Cisco IOS is affected by the following vulnerabilities:

- Processing ClientHello messages, documented as Cisco bug ID CSCsb12598
- Processing ChangeCipherSpec messages, documented as Cisco bug ID CSCsb40304
- Processing Finished messages, documented as Cisco bug ID CSCsd92405

Cisco has made free software available to address these vulnerabilities for affected customers. There are workarounds available to mitigate the effects of these vulnerabilities.

This advisory is posted at <http://www.cisco.com/warp/public/707/cisco-sa-20070522-SSL.shtml>.

**Note**

Another related advisory has been posted with this advisory. This additional advisory also describes a vulnerability related to cryptography that affects Cisco IOS. This related advisory is available at the following link:

<http://www.cisco.com/warp/public/707/cisco-sa-20070522-crypto.shtml>.

A combined software table for Cisco IOS is available to aid customers in choosing a software releases that fixes all security vulnerabilities published as of May 22, 2007. This software table is available at the following link:

<http://www.cisco.com/warp/public/707/cisco-sa-20070522-cry-bundle.shtml>.

**CSCsd85587**

A vulnerability has been discovered in a third party cryptographic library which is used by a number of Cisco products. This vulnerability may be triggered when a malformed Abstract Syntax Notation One (ASN.1) object is parsed. Due to the nature of the vulnerability it may be possible, in some cases, to trigger this vulnerability without a valid certificate or valid application-layer credentials (such as a valid username or password).

Successful repeated exploitation of any of these vulnerabilities may lead to a sustained Denial-of-Service (DoS); however, vulnerabilities are not known to compromise either the confidentiality or integrity of the data or the device. These vulnerabilities are not believed to allow an attacker will not be able to decrypt any previously encrypted information.

The vulnerable cryptographic library is used in the following Cisco products:

- Cisco IOS, documented as Cisco bug ID CSCsd85587
- Cisco IOS XR, documented as Cisco bug ID CSCsg41084
- Cisco PIX and ASA Security Appliances, documented as Cisco bug ID CSCse91999
- Cisco Unified Communications Manager, documented as Cisco bug ID CSCsg44348
- Cisco Firewall Service Module (FWSM)

This vulnerability is also being tracked by CERT/CC as VU#754281.

Cisco has made free software available to address this vulnerability for affected customers. There are no workarounds available to mitigate the effects of the vulnerability.

This advisory is posted at <http://www.cisco.com/warp/public/707/cisco-sa-20070522-crypto.shtml>.

**Note**

Another related advisory is posted together with this Advisory. It also describes vulnerabilities related to cryptography that affect Cisco IOS. A combined software table for Cisco IOS only is available at <http://www.cisco.com/warp/public/707/cisco-sa-20070522-cry-bundle.shtml> and can be used to choose a software release which fixes all security vulnerabilities published as of May 22, 2007. The related advisory is published at

<http://www.cisco.com/warp/public/707/cisco-sa-20070522-SSL.shtml>.

**CSCsd92405**

Cisco IOS device may crash while processing malformed Secure Sockets Layer (SSL) packets. In order to trigger these vulnerabilities, a malicious client must send malformed packets during the SSL protocol exchange with the vulnerable device.

Successful repeated exploitation of any of these vulnerabilities may lead to a sustained Denial-of-Service (DoS); however, vulnerabilities are not known to compromise either the confidentiality or integrity of the data or the device. These vulnerabilities are not believed to allow an attacker will not be able to decrypt any previously encrypted information.

Cisco IOS is affected by the following vulnerabilities:

- Processing ClientHello messages, documented as Cisco bug ID CSCsb12598
- Processing ChangeCipherSpec messages, documented as Cisco bug ID CSCsb40304
- Processing Finished messages, documented as Cisco bug ID CSCsd92405

Cisco has made free software available to address these vulnerabilities for affected customers. There are workarounds available to mitigate the effects of these vulnerabilities.

This advisory is posted at <http://www.cisco.com/warp/public/707/cisco-sa-20070522-SSL.shtml>.



#### Note

Another related advisory has been posted with this advisory. This additional advisory also describes a vulnerability related to cryptography that affects Cisco IOS. This related advisory is available at the following link:

<http://www.cisco.com/warp/public/707/cisco-sa-20070522-crypto.shtml>.

A combined software table for Cisco IOS is available to aid customers in choosing a software releases that fixes all security vulnerabilities published as of May 22, 2007. This software table is available at the following link: <http://www.cisco.com/warp/public/707/cisco-sa-20070522-cry-bundle.shtml>

CSCek48162: TDM cross connects before last call disconnect and assertions

**Symptom** : Under heavy stress few tdm assertion failures are seen

**Conditions** : This is seen with SS7 with more than 50 calls per second.

**Workaround**: There is no workaround

CSCek51075: Assertion failures at `tdm_local_endpoints_connect` CSCek61570 Trunk dn stuck in seize/seize state and does not recover.

**Symptom** : Few assertions may be seen during bootup and for the first set of calls. This does not have any effect on the system.

**Conditions** : This may happen in a situation when the calls are cleared as the system goes for a **rommon**.

**Workaround** : There is no workaround

CSCsb25337: Unnecessary tcp ports opened in default router config Cisco devices running IOS that support voice and are not configured for Session Initiated Protocol (SIP), are vulnerable to a crash. However, these devices are isolated to traffic destined to User Datagram Protocol (UDP) 5060. Devices which are properly configured for SIP processing are not vulnerable to this issue.

**Workaround** : See the advisory posted at:

<http://www.cisco.com/warp/public/707/cisco-sa-20070131-sip.shtml>

CSCsc72722: CBAC-firewall resets TCP idle timer upon receiving invalid TCP packets

**Symptom** : TCP connections that are opened through a Cisco IOS Firewall (CBAC) may not timeout.

**Conditions** : With Cisco IOS Firewall (CBAC) enabled, the TCP idle timer for a session may be reset even by TCP packets that fail TCP inspection and are subsequently dropped. This could lead to the TCP session not timing out.

**Workaround** : There is no workaround.

CSCsd91454: One way voice traffic due to incorrect **IPHC(UDP) Di0: CS 1 IPCRC**

**Symptom** : Voice traffic is dropped in one direction due to IPHC IPCRC error.

**Conditions** : The problem is found some time after the voice call has been established. When the problem is occurring, the logs show IPHC error messages.

**Workaround** : Use process switching

CSCsd92405: Router crashes on receipt of repeated SSL connection with malformed finished message

**Symptom** : A router crashes when receiving multiple malformed TLS and/or SSL3 finished messages. A valid user name and password are not required for the crash to occur.

**Conditions** : This symptom is observed when a router has HTTP secure server enabled and has an open, unprotected HTTP port.

**Workaround** : There is no workaround, however, user can minimize the chances of the symptom occurring by permitting only legitimate hosts to access HTTP on the router.

CSCse58397: ISDN BRI Dialer Interface is always in up state

**Symptom** : ISDN B channels are in UP state

**Conditions** : After reload and after shut/no shut

**Workaround** : There is no workaround

CSCsf28515: Crashes at mars\_default\_port\_dsp\_connect

**Symptom** : Router crashes at mars\_default\_port\_dsp\_connect after call passes through the digital voice-port.

**Workaround** : There is no workaround

CSCsf28711: 5850 reloads unexpectedly on making a single call CSCsf28840 crash due to configured peer type control vector

**Symptom** : Active eRSC reloads with traceback when first (PRI/SS7) call is made.

**Conditions** : This issue is seen when 5850tb is working with 12.4(10.5)PI5 image. Gateway come up with this image, when first (PRI/SS7) call is made the active eRSC reloads unexpectedly with traceback. This reload is seen for both H323 and SIP calls. Similar issue is seen in 5400 when MGCP-SIP call is made.

**Workaround** : There is no workaround

CSCsg16908: IOS FTP Server Deprecation

CSCsg46546: Erroneous alerting during pickup with CSCek58324. Call focus is wrong after picking up a trunk dn

**Symptom** : After an attempt to pick up an onhold trunk dn, the call display on the ephone which puts this DN to onhold is messed up. The call can not be picked up successfully by other phone and it becomes the focus one on the phone. The connected trunk dn can not be displayed and other incoming call can not be put on hold.

**Conditions** : There are two incoming trunk DN calls. The 1st one is answered and then the 2nd one. The 1st one is put onhold automatically when the 2nd one is answered. After the other phone attempts to pick up the 1st call, the pickup fails and the 1st call becomes the focus one on the phone. The softkey is displayed incorrectly.

**Workaround** : Press the line button to resume the call onhold instead of picking it up from pickup button or fac dialing. However, this workaround can not be applied to a phone which does not have the trunk DN configured.

CSCsg47834: NACK is observed for Open Voice Channel command

**Symptom** : NACK message may be received from 5510 DSP in response to Open Voice Channel command sent by the IOS.

**Conditions** : This problem may be observed when a same 5510 DSP is used as a Trans coding and Voice Termination resource.

**Workaround** : 1) Disable Trans coding (or)

2) Make sure that the Trans coding and Voice Termination are on different DSP(s). This can be performed by configuring the maximum number of trans coding sessions to a value such that it would require a multiple of 240 DSP credits. Example 1:

In the following configuration each trans coding session (complexity=high) will require 40 DSP credits. In order to use a multiple of 240 credits, we need to set the maximum trans coding sessions to 6 ( $6 * 40 = 240$ ) or any multiple of 6.

```
dspfarm profile 1 trans code
  codec g711ulaw
  codec g729r8
  associate application SCCP
Router(conf-t)#dspfarm profile 1 transcode
Router(config-dspfarm-profile)#maximum sessions 6
```

Example 2:

In the following configuration each transcoding session (complexity=medium) will require 30 DSP credits. In order to use a multiple of 240 credits, we need to set the maximum trans coding sessions to 8 ( $8 * 30 = 240$ ) or any multiple of 8.

```
dspfarm profile 2 trans code
  codec g711ulaw
  codec g711alaw
  codec g729ar8
  codec g729abr8
  associate application SCCP
Router(conf-t)#dspfarm profile 2 transcode
Router(config-dspfarm-profile)#maximum sessions 8
Use "show voice dsp group all" command to verify DSP resource allocation.
```

Note: Each 5510 DSP has 240 Credits. This work-around cannot be implemented if the router has only one PVDM2-16 which has only one DSP.

CSCsg59037: 851/871 cannot upgrade rommon from IOS

**Symptom** : Cisco 851 and 871 routers have no way to remotely upgrade the ROMMON firmware image.

**Conditions** : Cisco IOS versions for the Cisco 851 and 871 routers did not provide a mechanism to remotely upgrade the ROMMON firmware image.

**Workaround** : Cisco IOS Release 12.4(11)T1 for the Cisco 851 and 871 router introduces the command upgrade rom-monitor file which allows the ROMMON firmware image to be remotely upgraded. See this link for more information:

[http://www.cisco.com/univercd/cc/td/doc/product/software/ios124/124tcr/tcf\\_r/cf\\_13ht.htm#wp1032550](http://www.cisco.com/univercd/cc/td/doc/product/software/ios124/124tcr/tcf_r/cf_13ht.htm#wp1032550)

CSCsg66096: Privacy ON: call onhold can be intercepted by directed pickup operation

CSCsg66846: TNP phones opening new call when selecting shared transferring line

CSCsg68199: Trunk DN offhook is not propagated to a phone already in dial out mode

**Symptom** : Two IP Phones A and B are registered with Cisco Unified Communications Manager Express; these phones share two trunk DN's 1 & 2. If Phone-A goes offhook on DN-1 and Phone-B immediately goes offhook on DN-2. This condition should show the DN-2 button on Phone-A as busy which is not happening.

**Conditions** : This happens only when trunk DN's are used and they go offhook in quick succession on different phones and are in dialing mode.

**Workaround** : There is no workaround

CSCsg68711: Incoming call in background does not ring after transfer commit

**Symptom** : Phone does not ring for the second incoming call after committing transfer at alert for the first call.

**Conditions** : While transferring a trunk DN call, a call comes in. After committing the transfer at alert, the incoming call still does not ring on the phone.

**Workaround** : There is no workaround.

CSCsg70221: DTMF through the hairpin of a trunk DN does not work

**Symptom** : DTMF tones are being suppressed to prevent duplicate DTMF tones from being extended to an SCCP controlled VG224 port. This problem is a direct result of a fix implemented for correct CSCsf98754. The lack of DTMF prevents IVR devices from working correctly.

**Conditions** : PSTN -- FXO --- CME GATEWAY --- VG224/FXS --- IVR A call comes into a FXO port that is part of a trunk group and gets transferred to an extension that is hanging off of a VG224. DTMF is not relayed to the end point

**Workaround** : Setting the transfer system to full blind will prevent the DTMF blocking.

CSCsg70355: New default day light savings summer-time rules from Energy Policy Act of 2005 may cause Cisco IOS to generate timestamps that are off by one hour

**Symptom** : Starting in the calendar year 2007, daylight savings summer-time rules may cause Cisco IOS to generate timestamps (such as in syslog messages) that are off by one hour.

**Conditions** : The Cisco IOS configuration command: clock summer-time zone recurring uses United States standards for daylight savings time rules by default. The Energy Policy Act of 2005 (H.R.6.ENR), Section 110 changes the start date from the first Sunday of April to the second Sunday of March. It changes the end date from the last Sunday of October to the first Sunday of November.

**Workaround** : A workaround is possible by using the clock summer-time configuration command to manually configure the proper start date and end date for daylight savings time. For example: After the summer-time period for the calendar year 2006 is over, one can configure:

**clock summer-time PDT**

**recurring 2 Sun Mar 2:00 1 Sun Nov 2:00** (This example is for the US/Pacific time zone.)

CSCsg75035: Async Interface not showing up in the IfIndex from a remote NMS machine

**Symptom** : The interface is indexed on the router but the snmpwalk/snmpget keywords do not seem to return the value when the **sh snmp mib ifmib ifindex** command is used.

**Conditions** : This happens when loading a 3825 running **3825-adventerprisek9-mz.124-4.XC5.bin**

**Workaround** : There is no workaround

## Open Caveats - Cisco IOS Release 12.4(4)XC6

There are no known open caveats in this release.

## Resolved Caveats - Cisco IOS Release 12.4(4)XC5

CSCse56800

Multiple vulnerabilities exist in the Session Initiation Protocol (SIP) implementation in Cisco IOS that can be exploited remotely to trigger a memory leak or to cause a reload of the Cisco IOS device.

Cisco has released free software updates that address these vulnerabilities. Fixed Cisco IOS software listed in the Software Versions and Fixes section contains fixes for all vulnerabilities addressed in this advisory.

There are no workarounds available to mitigate the effects of any of the vulnerabilities apart from disabling the protocol or feature itself, if administrators do not require the Cisco IOS device to provide voice over IP services.

This advisory is posted at <http://www.cisco.com/warp/public/707/cisco-sa-20080924-sip.shtml>.

## CSCsf04754

Multiple Cisco products contain either of two authentication vulnerabilities in the Simple Network Management Protocol version 3 (SNMPv3) feature. These vulnerabilities can be exploited when processing a malformed SNMPv3 message. These vulnerabilities could allow the disclosure of network information or may enable an attacker to perform configuration changes to vulnerable devices. The SNMP server is an optional service that is disabled by default. Only SNMPv3 is impacted by these vulnerabilities. Workarounds are available for mitigating the impact of the vulnerabilities described in this document.

The United States Computer Emergency Response Team (US-CERT) has assigned Vulnerability Note VU#878044 to these vulnerabilities.

Common Vulnerabilities and Exposures (CVE) identifier CVE-2008-0960 has been assigned to these vulnerabilities.

This advisory will be posted at:

<http://www.cisco.com/warp/public/707/cisco-sa-20080610-snmpv3.shtml>

## CSCsf11855

Multiple voice-related vulnerabilities are identified in Cisco IOS software, one of which is also shared with Cisco Unified Communications Manager. These vulnerabilities pertain to the following protocols or features:

- Session Initiation Protocol (SIP)
- Media Gateway Control Protocol (MGCP)
- Signaling protocols H.323, H.254
- Real-time Transport Protocol (RTP)
- Facsimile reception

Cisco has made free software available to address these vulnerabilities for affected customers. Fixed Cisco IOS software listed in the Software Versions and Fixes section contains fixes for all vulnerabilities mentioned in this advisory.

There are no workarounds available to mitigate the effects of any of the vulnerabilities apart from disabling the protocol or feature itself.

This advisory is posted at

<http://www.cisco.com/warp/public/707/cisco-sa-20070808-IOS-voice.shtml>

## CSCse05642

Multiple voice-related vulnerabilities are identified in Cisco IOS software, one of which is also shared with Cisco Unified Communications Manager. These vulnerabilities pertain to the following protocols or features:

- Session Initiation Protocol (SIP)
- Media Gateway Control Protocol (MGCP)
- Signaling protocols H.323, H.254
- Real-time Transport Protocol (RTP)
- Facsimile reception

Cisco has made free software available to address these vulnerabilities for affected customers. Fixed Cisco IOS software listed in the Software Versions and Fixes section contains fixes for all vulnerabilities mentioned in this advisory.

There are no workarounds available to mitigate the effects of any of the vulnerabilities apart from disabling the protocol or feature itself.

This advisory is posted at

<http://www.cisco.com/warp/public/707/cisco-sa-20070808-IOS-voice.shtml>

CSCse68138

Multiple voice-related vulnerabilities are identified in Cisco IOS software, one of which is also shared with Cisco Unified Communications Manager. These vulnerabilities pertain to the following protocols or features:

- Session Initiation Protocol (SIP)
- Media Gateway Control Protocol (MGCP)
- Signaling protocols H.323, H.254
- Real-time Transport Protocol (RTP)
- Facsimile reception

Cisco has made free software available to address these vulnerabilities for affected customers. Fixed Cisco IOS software listed in the Software Versions and Fixes section contains fixes for all vulnerabilities mentioned in this advisory.

There are no workarounds available to mitigate the effects of any of the vulnerabilities apart from disabling the protocol or feature itself.

This advisory is posted at

<http://www.cisco.com/warp/public/707/cisco-sa-20070808-IOS-voice.shtml>

CSCek26492

**Symptom** : A router may crash if it receives a packet with a specific crafted IP option as detailed in Cisco Security Advisory: Crafted IP Option Vulnerability:

<http://www.cisco.com/warp/public/707/cisco-sa-20070124-crafted-ip-option.shtml>

**Conditions** : This DDTS resolves a symptom of CSCec71950. Cisco IOS with this specific DDTS are not at risk of crash if CSCec71950 has been resolved in the software.

**Workaround** : Cisco IOS versions with the fix for CSCec71950 are not at risk for this issue and no workaround is required. If CSCec71950 is not resolved, see the following Cisco Security Advisory: Crafted IP Option Vulnerability for workaround information:

<http://www.cisco.com/warp/public/707/cisco-sa-20070124-crafted-ip-option.shtml>

CSCsd40334: Processing a specially crafted IPv6 Type 0 Routing header can crash a device running Cisco IOS software. This vulnerability does not affect IPv6 Type 2 Routing header which is used in mobile IPv6. IPv6 is not enabled by default in Cisco IOS.

**Workaround** : Cisco has made free software available to address this vulnerability for affected customers. There are workarounds available to mitigate the effects of the vulnerability. The workaround depends on if Mobile IPv6 is used and what version on Cisco IOS is being currently used. See the advisory posted at: <http://www.cisco.com/warp/public/707/cisco-sa-20070124-IOS-IPv6.shtml>

CSCsd58381: Processing a specially crafted IPv6 Type 0 Routing header can crash a device running Cisco IOS software. This vulnerability does not affect IPv6 Type 2 Routing header which is used in mobile IPv6. IPv6 is not enabled by default in Cisco IOS.

**Workaround** Cisco has made free software available to address this vulnerability for affected customers. There are workarounds available to mitigate the effects of the vulnerability. The workaround depends on if Mobile IPv6 is used and what version on Cisco IOS is being currently used. See the advisory posted at: <http://www.cisco.com/warp/public/707/cisco-sa-20070124-IOS-IPv6.shtml>

CSCek56688: Change after-hours login timer to 1 min

**Symptom** : The minimum after-hours login timer is 5 minutes. It is too long. Customer wants to be able to deactivate the login in 1 min.

**Conditions** : The problem is observed when after-hours call blocking is enabled.

**Workaround** : There is no workaround.

CSCek58324: Call focus is wrong after picking up a trunk dn

**Symptom** : The call display does not work correctly when attempting to pick up an onhold trunk DN. The call cannot be picked up successfully by any other phone and it becomes the focus one on the single phone. The connected trunk DN cannot be displayed and other incoming calls cannot be put on hold.

**Conditions** : There are two incoming trunk DN calls. The first one is answered and then the second one. The first one is put onhold automatically when the second one is answered. After the other phone attempts to pick up the first call, the pickup fails and the first call becomes the focus on the single phone. The softkey is displayed incorrectly.

**Workaround** : Press the line button to resume the call onhold instead of picking it up from pickup button or fac dialing. However, this workaround cannot be applied to a phone that does not have its trunk configured for DN.

CSCsc74157: Pings fails with using ISDN switch-type primary-qsig

**Symptom** : A ping failed when using ISDN switch-type QSIG.

**Conditions** : This occurs with a Cisco 3725 and a Cisco 3845 back-to-back with ERNST-T2.

**Workaround** : There is no workaround.

CSCsd47303: Ephone template for ringing state

**Symptom** : With Cisco CME 4.0, an ephone-template has states for alerting, seized, connected and idle states. The softkey template needs to be defined for the ringing state (of an incoming call).

**Workaround** : There is no workaround.

CSCsd48251: Held call on shared line shows From Unknown Number

**Symptom** : After a certain amount of time, some calls that have been received on a shared line and placed on hold will show From Unknown Number.

**Workaround** : There is no workaround.

CSCse04642: CME GUI can not change ringtype for sidecar lines when log in as user

**Symptom** : When you log in as a user in Cisco Unified Communications Manager Express GUI, you cannot change the ringtype for sidecar lines. You can change the lines on the ip phone but not the lines that belong to the sidecar. If a user is logged in the Cisco Unified Communications Manager Express GUI (log in as user) and changes the ringtype via GUI for the sidecar line and then hits save, the action will save successfully but when you go to the line again the previous ringtype still shows.

**Conditions** : The problem is seen on Cisco IOS 12.3(14)T5 Cisco CME 3.3 and ios 12.4(4)XC1 and Cisco CME 4.0.

**Workaround** : This will work if the user changes from CLI to log in GUI as admin.

CSCse05642: I/O memory corruption crash on AS5850

**Symptom** : A redzone violation causes a Cisco AS5850 to crash.

**Conditions** : This symptom is observed on a Cisco AS5850 gateway having MGCP-NAS package and outgoing VoIP calls.

**Workaround** : There is no workaround.

CSCse56800: SIP-3-BADPAIR register timer expiry causes slow memory leak

**Symptom** : SIP Processes causing slow memory leak when there are no active calls on a Cisco 3725. Specifically, the SIP register timer expiry messages are causing this behavior. Reloading the router does not resolve the issue.

**Conditions** : The message below is what causes this behavior:

```
007042: Jun 17 15:18:45.024 EDT: %SIP-3-BADPAIR: Unexpected timer 23
(SIP_TIMER_REMOVE_TRANSACTION) in state 27 (SIP_STATE_OPTIONS_WAIT) substate 0
(SUBSTATE_NONE)
```

**Workaround** : There is no workaround

CSCse68138: Handle fragmented packets in VOIP RTP Lib

**Symptom** : Router may reload due to fragmented RTP packets. This is a platform independent problem.

**Conditions** : This problem is likely to happen in networks where VOIP is one of applications and one more segments of network are using low MTU.

**Workaround** : There is no workaround.

CSCse71162: Change minimum ephone keepalive timer from 10 to 1 second

**Symptom** : Request to reduce the minimum configurable keepalive timer from 10 to 1 second in Cisco Unified Communications Manager Express for SCCP phones.

**Workaround** : There is no workaround.

CSCse82300: Getting Undefined Tone when we enter a invalid FAC

**Symptom** : The CFA feature in the Cisco VG224 is enabled and we are dialing an invalid FAC code via callgen. We expect to get a reorder tone immediately but we are getting only the Undefined\_tone.

**Workaround** : There is no workaround.

CSCse83674: FXS port cannot be recovered when offhook with howler tone at end of call

**Symptom** : Analog FXS port on a Cisco 2800/3800 ISR does not go back to idle if it has been offhook for more than a minute at the end of a call.

**Conditions** : A and B are two FXS ports on the same router connected to analog phones. A calls B. B answers the call. Once the conversation is done, A hangs up. B does not go onhook. After 60 seconds, B starts hearing offhook alert (howler) tone. Putting B onhook now has no effect. B continues to play offhook alert for the rest of its life until the router is reloaded.

**Workaround** : There is no workaround.

CSCse87446: Extension assigner defaults provision-tags to 0

**Symptom** : Extension assigner will chose wrong extension if the provision-tag input is zero.

**Workaround** : Use the ephone-tag.

CSCsf02737: Memory Corruption Crash at chunk\_free\_caller

**Symptom** : A Cisco 3825 running Cisco IOS 12.4-9.T crashed. The decoded tracebacks is as follows:

```

abort
crashdump
chunk_free_caller
free_lite_internal
__free
free
skinny_send_msg_internal
skinny_server_process
r4k_process_dispatch

```

**Conditions** : This seems similar to CSCsb80447.

**Workaround** : Configuring **no memory lite** seems to alleviate the crashes.

CSCsf07990: CME Dynamic Hunt-Group Login fails

**Symptom** : Ephone-1 has extension 88, which is also added as a monitor line on a 7914. The Ephone-2, which is connected to the 7914 is in DND state. Now when you try to login to a hunt-group on ephone-1, it fails because the ephone with the monitor lines is in DND state.

```

Aug 14 08:36:07: SkinnyHGJoinByDn: dn(88), join_code(80), join(1)
Aug 14 08:36:07: Cannot join 88 to hunt group list with dnd on.
Aug 14 08:36:07: ephone-1[13]:SkinnyHGJoinByPhone phone-[7] join 80 failed.

```

**Workaround** : Ephone with the Cisco IP Phone 7914 should not be in DND state.

CSCsf21007: Ephone hunt-group does NOT present calls to monitored DNs

**Symptom** : When an ephone hunt-group is configured with **present-call idle-phone**, the ephone hunt-group skips over certain members of the hunt group.

**Conditions** : The problem is observed when members of the ephone hunt-group are monitored.

**Workaround** : Do not monitor the members of the hunt-group.

CSCsf21458: SRST Reuses sockets causing phones unregister

**Symptom** : Registered ephones in SRST mode may unregister and then re register

**Conditions** : This happens when the phone requests for a socket that has already been used by another ephone.

**Workaround** : There is no workaround.

CSCsf98754: Inband DTMF should be squelched for calls from POTS to Skinny

**Symptom** : The following scenario is seen:

PSTN === Analog or T1 CAS FXO === CME ----- VG224 ---- Phone or IVR

The analog ports on the Cisco VG224 are SCCP controlled by Cisco Unified Communications Manager Express.

For a call between PSTN and a Cisco VG224 port (or an IP Phone), the DTMF detection is turned ON on the FXO port. Along with this, the DSP channel associated with the FXO port is programmed to pass through the DTMF tone in the RTP path instead of suppressing it.

The above manifests into a double DTMF digit scenario and is very well pronounced when the Cisco VG224 port is connected to an IVR system looking for digits. For the endpoints controlled by Cisco Unified Communications Manager Express via SCCP, the DTMF relay happens through out of band SCCP messages. Since the original DTMF digit coming from PSTN is not suppressed, we see two digits reaching the IVR system - one from the SCCP message from Cisco Unified Communications Manager Express to the Cisco VG224 port and the second one embedded in the RTP path.

**Conditions** : A simple way to reproduce this problem is as follows:

Phone----FXS=CME----- IP Phone or VG224

Make a call from phone on the left to a Cisco Unified Communications Manager Express controlled endpoint. Press a digit button on the left phone and hold it for a long time. The user on the Cisco Unified Communications Manager Express controlled endpoint on the right can hear: digit beep, silence and continuous digit beep. If the squelching flag was set on the FXS DSP channel, the user would have heard digit beep, silence and back to voice path.

**Workaround** : There is no workaround.

CSCsf99737: SRST Locale fail over soft keys still display English

**Symptom** : SRST fails over from Cisco Unified Communications Manager still displays English languages in softkey regardless of the languages that is configured in Cisco Unified Communications Manager.

**Workaround** : There is no workaround.

## Open Caveats - Cisco IOS Release 12.4(4)XC5

There are no known open caveats for this release.

## Resolved Caveats - Cisco IOS Release 12.4(4)XC4

CSCse68355

Multiple voice-related vulnerabilities are identified in Cisco IOS software, one of which is also shared with Cisco Unified Communications Manager. These vulnerabilities pertain to the following protocols or features:

- Session Initiation Protocol (SIP)
- Media Gateway Control Protocol (MGCP)
- Signaling protocols H.323, H.254
- Real-time Transport Protocol (RTP)
- Facsimile reception

Cisco has made free software available to address these vulnerabilities for affected customers. Fixed Cisco IOS software listed in the Software Versions and Fixes section contains fixes for all vulnerabilities mentioned in this advisory.

There are no workarounds available to mitigate the effects of any of the vulnerabilities apart from disabling the protocol or feature itself.

This advisory is posted at

<http://www.cisco.com/warp/public/707/cisco-sa-20070808-IOS-voice.shtml>

CSCsc74783

**Symptom** : Intrusion Prevention System (IPS) signatures that require inspection of TCP flows below port 550 may not be triggered on a Cisco IOS IPS device.

**Conditions** : This symptom is observed on a Cisco IOS router that is configured for IPS functionality.

**Workaround** : Apply CBAC (Context Based Access Control) in addition to IPS.

**Further Information:** On a Cisco IOS router with IPS (Intrusion Prevention System) enabled, all TCP flows should be subject to TCP stateful inspection until the TCP 3-way handshake is complete. This does not work for TCP sessions with a destination port that is less than 550, if it does not match a predefined signature on the router.

CSCek47681: Backplane TDM loss and assertion failures

**Symptom** : Under heavy stress, time division backplane timeslots may be lost over time.

**Conditions** : The symptom occurs with SS7 and more than 50 calls per second.

**Workaround** : There is no workaround.

CSCse06975: Traceback at pak\_copy\_contiguous\_to\_contiguous when testing multicast

**Symptom** : The VoIP LMR multicast does not function properly with E&M on the NM-HD-2V network module.

**Workaround** : There is no workaround.

CSCse16973: **show controller call-counters** displays negative values

**Symptom** : The **show controller t1 call-counters** command displays negative values for the DSO **Active** counter.

**Conditions** : The symptom occurs on the Cisco AS5400XM platform for both voice and data calls.

**Workaround** : There is no workaround.

CSCse18940: Memory depletes when VoAAL2 traffic is passed.

**Workaround** : There is no workaround.

CSCse27845: One way voice after ringing pickup of transferred at-alert call

**Symptom** : The called party may not be able to hear the caller.

**Conditions** : Phones A, B, and C are controlled by the same Cisco Unified Communications Manager Express. A calls B. B does an at-alert transfer to C. While C is ringing, B does a ringing pickup on C's extension. One-way voice results with B being unable to hear A.

**Workaround** : There is no workaround.

CSCse47728: Path confirmation failures are observed with VoATM

**Symptom** : Path confirmation failures seen with Voice over ATM traffic.

**Conditions** : This is seen with only with VoAAL2 traffic.

**Workaround** : There is no workaround.

CSCse50167: Speed dial line buttons disappear from CME phones after the router reloads.

**Conditions** : The speed dials are configured using an ephone template, which is then applied to the affected phone.

**Workaround** : Remove and re-apply the ephone template after the router reloads.

CSCse56129: Cisco VG224 erroneously triggers hookflash during CME call pickup interaction

**Symptom** : On the Cisco VG224, a voice port registered to Unified Communications Manager Express running 12.4(4)XC may falsely detect a hookflash in the call pickup case.

**Conditions** : During call pickup, the Cisco Unified Communications Manager Express sends an onhook signal to the VG224 port, presents a new call and immediately instructs the port to move to connected state. During these quick steps, the voice port on the VG224 is erroneously reporting a hookflash.

**Workaround** : Configure **no supervisory disconnect lcfo** on the Cisco VG224 voice port to avoid the false hookflash detection in the Cisco Unified Communications Manager Express call pickup case.

CSCse56660: Inbound calls to fxo port fail (no audio) when caller-id enabled

**Symptom** : Inbound calls to Foreign Exchange Office (FXO) ports on Cisco IOS VoIP gateways connect, but audio is not present.

**Conditions** : With caller-id enable configured on FXO ports, the call will connect, but no audio is heard. When this occurs, the following error message can be seen at debug level:

```
Jun 20 01:41:15.855: mbrd_elt1_vic_connect: setup failed
Jun 20 01:41:15.855: flex_dsprmtdm_xconn: voice-port(0/0/1), dsp_channel
(/0/2/0)
```

**Workaround** : Disable caller-id on the voice port.

CSCse59347: CME/SRST IP phone unregister does not down the virtual pots peers

**Symptom** : Using SRST 4.0 with Cisco Unified Communications Manager Express, calls may fail with a “user busy” signal.

**Conditions** : When the IP phone must unregister/fall back to the Cisco Unified Communications Manager, the virtual POTS dial-peers do not disconnect and calls fail with user busy rather than being sent via the H.323 dial-peer to the Cisco Unified Communications Manager.

**Workaround** : There is no workaround.

CSCse69235: 871 XC - S&K interface forwarding results in hung interface

**Symptom** : VLAN interfaces on Cisco 870 series routers may cease to function under heavy loads.

**Conditions** : If the 802.1x feature is configured as a layer 3 transport in 12.4(4)XC images and continuous, heavy, and unauthenticated traffic is received on a virtual interface, the router may stop responding.

**Workaround** : There is no workaround.

CSCse70333: CFwdAll erroneously reconfigured after disabling night service

**Symptom** : **CFwdAll** incorrectly appears after night service is disabled.

**Conditions** : **CFwdAll** was initially configured using softkey, and unconfigured through the CLI. On the same DN as CFwdAll was on, night service is enabled and disabled.

**Workaround** : Remove **CFwdAll** via softkey or reload the router.

CSCsc42589: Reset msg to TAPI client when phone reset restart by CME.

CSCsc72502: The TAPI client may not show the call lines in ringing or connected state for the controlled ephone.

**Conditions** : If the TAPI client registers to the Cisco Unified Communications Manager Express while its controlled ephone has some connected or ringing lines, it would not show their status. It would show them all in IDLE state. This problem occurs in any Cisco Unified Communications Manager Express releases.

**Workaround** : There is no workaround.

CSCse06975: Traceback at pak\_copy\_contiguous\_to\_contiguous when testing multicast

**Symptom** : VoIP LMR multicast capability does not work on network module NM-HD-2V with E&M.

**Workaround** : There is no work around.

CSCse15025:Intermittent analog/cas voice port lockup or robotic voice

**Symptom** : An analog or digital CAS port enters a state in which inbound or outbound calls, or both, may no longer function through the port.

**Conditions** : This symptom is observed on a Cisco 2800 series and Cisco 3800 series that function as gateways with analog or digital CAS ports that use PVDM2 DSP modules.

When this problem occurs, it impacts multiple ports that share the same signaling DSP. The output of the **show voice dsp signaling** EXEC command shows which DSP is used by a port for signaling. The symptom may occur more often for ports that use DSP 1 on the PVDM2 module for signaling.

Because this issue impacts the signaling channels, it has been seen that calls either will not connect at all through impacted ports or in some cases when multiple simultaneous calls are present on adjacent voice ports/timeslots, the call may connect momentarily before being disconnected.

If a problem occurs only on a single voice port, there is another problem, not this caveat (CSCse15025). PRI/BRI calls are not affected because PRI/BRI does not utilize the DSP for signaling purposes.

When the symptom occurs with either a VIC2-xFXO or EVM DID/FXS module, enter the **terminal monitor** command followed by the **test voice port port-number si-reg-read 39 1** command for one of the affected ports. The output typically should be a single octet value for register 39. When the symptom occurs, information for Registers 40, 41, and 42 is presented and some of the registers show double- octet information. See the example output (2) below.

When the symptom occurs with FXS or analog E&M modules, enter the **terminal monitor** command followed by the **test voice port port-number codec-debug 10 1** command for one of the affected ports. The output typically should be a single octet value for each register. See the example output (4) below.

**Workaround** : There is no workaround.

CSCse47338: H245-signal dtmf relay requires signal update to end digits

**Symptom** : A third party device sends dtmf-relay using a h.245-signal, which includes duration of the digit. The Cisco Unified Communications Manager Express gateway sends the digit to CUE, but the digit is not considered done unless another digit is received. This results in %SIP-3-DIGITEND: Missing digit end event messages sages.

**Workaround** : Send an extra (unnecessary) digit, which indicates the previous digit is ended.

CSCse60250: Support Localization for the Cisco IP Phone 7906 on Cisco Unified CME.

CSCse66125: Call-waiting ring in ephone-dn-template fails to hold configuration

**Symptom** : When trying to configure call-waiting ring on an ephone-dn x, the configuration is accepted, but cannot be seen in the configuration.

CSCse75014: CME/SRST not able to make calls to Unity VM

**Symptom** : With Cisco Unified Communications Manager Express/SRST, you are able to make calls to Unity VM. VM port DN is not coming to “Idle” state after restarting Unity.

CSCeh69448: SCCP CME need to clean up tftp binding.

CSCek43094: Add TNP compatible Network locale tags to cnf file.

CSCsc82351: Device ID for the Goped phone is incorrect

**Symptom** : The device ID for the Goped phone is incorrect.

**Workaround** : There is no workaround.

CSCsc85575: Subsequent call following a conf call by TNP Ph results in 1-way audio

**Symptom** : No audio is received from a Cisco 7931 IP phone.

**Conditions** : This symptom is observed when a call is made between a Cisco IP phone 7960 and a Cisco IP phone 7931. The user of the CiscoIP phone 7960 experiences one-way audio intermittently while the user of the Cisco IP phone 7931 does not experience this symptom.

**Workaround** : Reset the Cisco IP phone 7931.

CSCsc99639: CME unable to make call on 2nd line using line button when 1 line busy

**Symptom** : The CME is unable to make call on a second line using line button when line 1 is busy

**Conditions** : This occurs when you make a call from Phone A to Phone B on Line 1. Answer the call on Phone B on line 1. Press Line 2 on Phone B. The first call is put on Hold on Line 1 but Line 2 button light does not come up and Line 2 has no dial tone and it does not accept a new call on Line 2 at all. Ideally Line 2 should put the call on Hold and then accept new call with giving out dial tone.

**Workaround** : There is no workaround.

CSCsd13066: No caller ID displayed for a forwarded call on IP Phone running 7.x

**Symptom** : When release 7.x phoneload is used on a forwarding phone, the forward-to party does not see the forwarded party number on the display.

**Workaround** : There is no workaround.

CSCsd73435: The **button-layout help** CLI is unclear.

CSCsd86966: Not able to create CTL file for 7906 phone.

CSCsd90419: Cisco IP Phone 7941/61/11 does not support localization in SRST

**Symptom** : The Cisco 7941/61/11 phones display change to English in SRST mode.

**Conditions** : Phone falls back to SRST Cisco Unified Communications Manager Express router.

**Workaround** : There is now workaround.

CSCse05698: CME 12 build in locales support on 7941/61/11.

CSCse08865: Enhance CME locale installer to support 7941/61/11/70/71

CSCse16210: 7920 locale support enhancement.

CSCse29308: CCME extension assigner extra

CSCse35293: CCME extension assigner need to update CNF file.

CSCse36127: If a Phone is viewed on the GUI the extensions are marked as normal ring even if they are monitored lines. So every time a change is made all lines have to be corrected via the CLI.

**Workaround** : This defect has been rectified via the Cisco Unified Communications Manager Express GUI 4.0.0.1a file package. Download and install this Cisco Unified Communications Manager Express GUI file package (or newer) to overcome the problem.

CSCse39419: Some phones XML file does not have correct m\_vendor

**Symptom** : Cannot configure the phone through the vendorConfig in the XML file

Further Problem Description: The VendorConfig is missing in the XML file.

**Workaround** : There is no workaround.

CSCse41295: MOH debugs flood the console when MOH file is unconfigured

CSCse56023: CME extension assigner clean up

CSCse62649: Change CME GUI logo to Cisco Unified Communications Manager Express

CSCse65819: Reset needed after extension assignment of 7914 attached phone

## Open Caveats - Cisco IOS Release 12.4(4)XC4

There are no known open caveats for this release.

## Open Caveats - Cisco IOS Release 12.4(4)XC4

There are no known caveats for this release

## Resolved Caveats - Cisco IOS Release 12.4(4)XC3

CSCek37177: The Cisco IOS Transmission Control Protocol (TCP) listener in certain versions of Cisco IOS software is vulnerable to a remotely-exploitable memory leak that may lead to a denial of service condition.

**Symptom** : This vulnerability only applies to traffic destined to the Cisco IOS device. Traffic transiting the Cisco IOS device will not trigger this vulnerability.

**Workaround** : Cisco has made free software available to address this vulnerability for affected customers. There are workarounds available to mitigate the effects of the vulnerability. See the advisory posted at; <http://www.cisco.com/warp/public/707/cisco-sa-20070124-crafted-tcp.shtml>

CSCek38136: Hissing noise heard before ringback tone begins.

**Symptom** : When you deploy VoIP using PVDM2 / 5510 DSP modules, a hissing sound may be heard before the ringback tone starts on the calling side.

**Conditions** : This symptom is observed only with 5510 DSP modules. The symptom does not occur with 549 DSP modules.

**Workaround** : There is no workaround.

CSCin96828:On TCW, calls may be reject with the "Duplicate Setup: error message.

**Symptom** : Alternate call retry times out because of no response from the terminating gateway.

**Conditions** : Both originating gateway and terminating gateway are registered to the gatekeeper, and gatekeeper returns the same physical gateway as both primary endpoint and alternate endpoint with a different DNIS. If the terminating gateway receives the alternate call before the primary call is cleaned up, the terminating gateway considers the alternate call as duplicate and ignores the SETUP.

**Workaround** : When same gateway has to be used as primary and alternate, use two different address of the gateway instead same address.

CSCsd13074: Hold-alert shared-and-idle enhancement is needed.

CSCse44965: Cisco Unity does not register to CME using the **no auto-reg-ephone** command. Cisco Unity does not register to Cisco Unified Communications Manager Express while using the following command: no auto-reg-ephone

**Symptom** : Cisco Unity does not register with Cisco Unified Communications Manager Express 4.0

**Conditions** : This symptom occurs when the command **no auto-reg-ephone** is configured for telephony-service.

**Workaround** : Enable **auto-reg-ephone** for the telephony -service.

CSCse37580: Router crashes using the **@ppc\_process\_dispatch** command.

**Symptom** : A Cisco router crashes while making a call using the command **test voice port detector ring-trip**.

**Workaround** : There is no workaround.

CSCse41306: The sound quality of the Music on Hold feed is poor.

**Workaround** : Using ST-TC1 in tandem with TBB-1 may work in some cases.

CSCse23652: 7912 registered to CME 3.3 running Cisco IOS 12.4(3b) is not able to initiate a call when controlled by the TSP client running TSPv.

**Workaround** : There is no workaround.

CSCse23304: When selecting "Extensions" in the Configuration drop-down menu in the Cisco Unified Communications Manager Express GUI for CME 4.0, extension type "Park-Slot" is erroneously listed as being of the type "MWI" and extensions of the type "Paging" are erroneously listed as being of the type "MOH." All other extension types are listed correctly.

**Conditions** : The symptom occurs on Cisco Unified Communications Manager Express 4.0 voice routers using Cisco Unified Communications Manager Express GUI version 4.0.0.0.

**Workaround** : If Cisco Unified Communications Manager Express 4.0 features are not needed, use an earlier release of Cisco Unified Communications Manager Express and its corresponding GUI version. If the Cisco Unified Communications Manager Express 4.0 features are needed, configure **ephone-dns** and ephones via Cisco IOS CLI.

CSCse20435: Memory runs out and fails the system after extended use.

**Conditions** : The memory usage increases when an ephone falls back from a Cisco Unified Communications Manager Express or CCME to a SRST.

**Workaround** : There is no workaround.

CSCsd99389: The system reloads when call forwarding is invoked.

**Conditions** : The symptom occurs if a dialplan pattern and an invalid call forward number are configured.

**Workaround** : Verify that all call forward numbers are correct or remove the dialplan pattern.

CSCsd57413: Unhide the 7906G-related CLI.

CSCsd46996: A memory leak occurs while upgrading CAPF

**Workaround** : There is no workaround.

CSCek45370: A dialer interface using VWIC-MFT-2T1/E1 may get an ISDN carrier timeout if the terminating gateway does not have dialer-group configured.

**Workaround** : Configure dialer-group on the terminating gateway.

CSCek40644: One channel of a DN may get stuck and incoming and outgoing calls cannot be made.

**Workaround** : Reset the ephone.

CSCsd19564: Button optimization is required for park-recall, pickup-on-hold and pickup-at-alert functions.

CSCsd57096: Configuration is prohibited for certain interfaces through the interface range command.

**Conditions** : This symptom occurs on all platforms if the interface range command is used to attempt to configure a range of interfaces when the last physical interface in the range itself contains the sub-interfaces.

**Workaround** : Configure each of the interfaces individually and outside of the interface range command.

Another option is to remove the subinterfaces from the last interface in the range prior to attempting the configuration through the range command.

CSCse19112: Caller name does not display on all IP phones that have shared lines or overlay lines.

**Workaround** : There is not workaround.

CSCse35506: Secondary dialtone is not heard after FAC standard is configured on CME 4.0

**Workaround** : Call can be placed without the dial tone.

CSCse34614: Caller name does not appear on IP phone display.

**Conditions** : The symptom occurs with IP phones that are registered to CME with IOS 12.4(4)XC (CME 4.0), with IP phones that have overlay DNs with call waiting, and with IP phone that has an active call on the overlay dn and a second call is received.

**Workaround** : There is no workaround.

CSCeg90328: The router crashes when the user tries to convert an sccp phone to an SIP phone.

CSCek34261: A Cisco Integrated SONET/SDH router (ISR) may crash during the "gt96k\_mbrd\_bri\_set\_bandwidth" function.

**Conditions** : This symptom is observed on the Cisco 1800, 2800, and 3800 series routers that function as an ISR when an incoming call is placed with 32KB bandwidth. The symptom does not occur when has a call has 56 KB or 64 KB bandwidth.

**Workaround** : Deny the invalid incoming call by entering the `<CmdBold>isdn caller</noCmdBold>` command on the ISR router.

CSCse39049: Cisco Unified Communications Manager does not show any output for "show call-manager-fallback ephone-d."

**Workaround** : There is no workaround.

## Open Caveats - Cisco IOS Release 12.4(4)XC3

There are no known open caveats for this release.

## Resolved Caveats - Cisco IOS Release 12.4(4)XC2

CSCsd54232: Qsig Call Forward shows AFW memories leak

**Symptom** : Certain QSIG Call Forward Busy and Call Transfer operations might result in memory leakage.

**Workaround** : There is no workaround.

CSCsd78806: Can not add ATA through GUI

**Symptom** : While trying to add an ATA through the GUI for Cisco Unified Communications Manager Express, an error is received on the page when you try to save the ATA configuration information and its not possible to add the ATA.

**Conditions** : This applies to only ATA; when you try to add any other IP phone, the process works fine.

**Workaround** : Add ATA through the CLI on the router.

CSCsd35389: Deleted ephone-dn's never unregister with Gatekeeper unless GW reloaded

**Symptom** : Usually CME is automatically registered with a **gatekeeper all** for the ephone-dn, but when a ephone-dn is deleted it never unregisters with the gatekeeper.

**Conditions** : This problem occurs even if you enter **no gateway** then **gateway** on the CME router to force it to unregister.

**Workaround** : Reregister the deleted ephone-dn and enter **show back up**.

CSCsd08105: Cisco IP Phone 7960 speaker phone light does not turn off after pressing hold

**Symptom** : When a call is put on hold on a Cisco IP Phone 7960 and the call is on a speaker phone, resuming on another phone does not clear the speakerphone light (release 7.x phoneload).

**Workaround** : Depress the speakerphone light to clear it.

CSCsd71081: CME 4 Crashing & Tracebacks when DND button pressed on 7970 with 7914

**Symptom** : Cisco Unified Communications Manager Express 4 Reloads after DND button is pressed.

**Conditions** : This problem occurs when using the Cisco IP Phone 7970 with the additional 7914 extension module; pressing the DND button causes the Cisco Unified Communications Manager Express to produce traceback and reload. This does not occur when the 7914 is not installed.

**Workaround** : There is no workaround.

CSCsc11833: Intermittent analog or digital CAS port lock up - in/out calls fail

**Symptom** : An analog or digital CAS port gets into a state where inbound and/or outbound calls through the port may no longer work.

**Conditions** : This symptom has been seen on Cisco 2800/3800 gateways with analog or digital CAS ports which use PVDM2 DSP modules.

It can take some time for the symptom to occur, but when it does occur, it impacts multiple ports which share the same signaling DSP. To see which DSP a port is using for signaling, check the output of the exec command **show voice dsp signaling**. It has been observed to occur more often with those ports which use DSP 1 on the PVDM2 module for signaling.

If a problem is noticed only on a single voice port, it would not be this issue. Because PRI/BRI does not utilize the DSP for signaling purposes, it is not impacted by this issue.

When the problem occurs and this is either on a VIC2-xFXO or EVM DID/FXS modules, run **test voice port # si-reg-read 39 1** on one of the impacted ports. You need to run 'terminal monitor' first to see the output. The output typically should be a single octet value for register 39. When the problem happens, information for Registers 40, 41 and 42 is presented as well and some of the registers show double-octet information.

If using FXS or analog E&M modules, use **test voice port # codec-debug 10 1** and compare the output. Again, the normal output will be single octet information for each register. This test only needs to be run on one of the voice ports in this state to confirm if this is the issue being seen.

**Workaround** : There is no workaround to prevent this problem from occurring. Once in this state, a reload of the gateway is necessary to recover it.

Additional Information: If the problem being seen has been confirmed to be this issue, the software changes associated with this report will mitigate the problem in the majority of cases. It may still be possible to see the problem in some cases and if this is experienced contact Cisco TAC for assistance.

CSCsd71944: No audio on consult transfer to PSTN with live MOH for local users

**Symptom** : This was a bug in the multicast MoH for local SCCP phones features. Multicast MoH was not stopped for the transferee (if it was an SCCP phone) when the transfer was committed.

To check whether you are experiencing this bug, look for debug output similar to the following after the transfer is committed:

```
*Mar 20 23:37:04.795: SkinnyUpdateDnState by EFXS_OPEN_VOICE_PATH
  for DN 1 chan 1 to state CALL_START
*Mar 20 23:37:04.795: ephone-1[3]:UpdateCallState DN 1 chan 1 state 12 calleddn
-1 chan 1
*Mar 20 23:37:04.795: ephone-1[3]:Binding ephone-1 to DN 1 chan 1 s2s:0
*Mar 20 23:37:04.795: SKINNY_CALL_START for DN 1 has call_restart set
*Mar 20 23:37:04.795: SKINNY_CALL_START for DN 1 change to CONNECTED
*Mar 20 23:37:04.795: ephone-1[3]:Call Start ignored - mediaActive set
```

**Conditions** : The transferee is an SCCP phone listening to multicast MoH supplied by the Cisco Unified Communications Manager Express router.

**Workaround** : A simple workaround is to remove the “multicast moh...” configuration under “telephony-service.”

CSCsd76246: Intercom not muting 7960s

**Symptom** : Cisco IP Phone 7960s are not becoming muting when receiving intercom calls.

**Conditions** : A Cisco IP Phone 7960 is configured in an intercom dn pair with another phone on the same system.

**Workaround** : There is no workaround.

CSCsd85687: Secondary number is not considered while parking to reserved-for slot

**Workaround** : There is no workaround.

CSCsc46528: Sub entry polling: no of rows returned inconsistently from speedbird mib

**Symptom** : ccmeEphoneActTable from CISCO-CCME-MIB provides inconsistent Vresults.

**Conditions** : This symptom has been observed when a partial SNMP GET is issued on selected columns from ccmeEphoneActTable.

**Workaround** : Perform a complete SNMP GET instead of a few entries on ccmeEphoneActTable.

CSCsd91169: Hlog from ringing state allows last agent to logout of b-acd

**Workaround** : There is no workaround.

CSCsd27683: Cisco IOS gateway does not make H.245 TCP connection after it gets the address

**Symptom** : A Cisco IOS H.323 gateway (GW) that is running Cisco IOS Release 12.4 (7) is not initiating the H.245 TCP connection.

**Conditions** : This symptom occurs only if the terminating GW or CCM sends Alert with H.245 Address and PI=1,2,8 in response to a fastStart Setup sent from the originating GW.

**Workaround** : Perform the following:

- Add **progress\_ind alert strip** on outgoing dial peer to TGW in OGW.
- Configure slow start on the gateways. Under voice service VoIP, H.323 mode.

Further Problem Description: An H.323 GW initiates the FS call to another GW or CCM. In response to this, CCM or terminating GW sends slow start Alert with h245 Addr and PI=1,2,8. The phone at the originating GW expects ringing tone from the terminating GW. It is not ringing now, but the phone at the terminating side is ringing. Now if user did not pick the call (i.e. will not send Connect message), then the call will drop. Caller will never come to know what happened at the other end (there is no ringing tone). Without PI in Alert, it works well.

CSCsc94215: Fix indexing in various tables in CISCO-SRST-MIB & related components

**Symptom** : The index may not come out correctly if doing a get next of an item in a table.

**Conditions** : This occurs in csrstAliasTable, csrstAccessCodeTable, csrstLimitDNTable, ccmeCorConfTable, ccmeDialplanPatternTable, and csrstSipEndpointTable.

**Workaround** : Use get bulk to get the complete table, which returns all correct values.

CSCek38822: no-reg doesnt work on SRST

**Workaround** : There is no workaround.

CSCek40136: the no-reg on ephone-dn should not apply to dialplan-pattern

**Symptom** : The E164 number cannot be registered if the matched extension in the dialplan-pattern has no-reg configured. The problem started to occur in 12.4(5.11).

**Conditions** : If the number in the ephone-dn is configured with no-reg, the expanded E164 via dialplan-pattern can't be registered to either Gatekeeper or SIP proxy.

**Workaround** : The extension needs to be registered in order to have the expanded E164 registered.

CSCsd96951: Call Pickup fails for 2nd channel on dual line ephone dn if 1st is busy

**Symptom** : Call cannot be picked up on 2nd channel of a dual line ephone-dn.

**Conditions** : Using Cisco Unified Communications Manager Express 4.0 (Cisco IOS Software 12.4.4XC1) and 1st channel of ephone-dn is BUSY.

**Workaround** : Use a different phone or different ephone-dn (button) on the same phone.

CSCsd9109: Cisco IP Phone 7936 registration blocked by CLI: **no auto-reg-ephone**

**Symptom** : Cisco IP Phone 7936 will not register with Cisco Unified Communications Manager Express:

%IPPHONE-6-REG\_ALARM:

**Conditions** : This problem occurs in Cisco IOS Software Release 12.4(4)XC. If **no auto-reg-ephone** is configured, then the user is unable to manually register the ephone associated with a Cisco IP Phone 7936.

## Open Caveats - Cisco IOS Release 12.4(4)XC2

There are no known open caveats in this release.

## Resolved Caveats - Cisco IOS Release 12.4(4)XC1

CSCed28266: Software forced crash when adding SIP call headers

**Symptom** : A Cisco gateway may unexpectedly reload because of a software-forced crash when it builds a SIP ACK(nowledge)ment or BYE message.

**Conditions** : This symptom is observed when the gateway receives a SIP response that contains a Record-Route header and a Contact header and when the length of the Contact header exceeds 128\*n, in which "n" is the number of URLs in the Record-route header.

**Workaround** : There is no workaround.

CSCej73716: Dot1x configurations on L2 and L3 ports should be mutually exclusive

CSCek25126: FXO trunk dn and monitoring enhancement for customer

CSCek26750: ogw(h323,slow start)--ipip1(sip)--ipip2(sip)--tgw not working

**Symptom** : Voice call fails in the following scenario ogw (h323, slow start) --  
ipipgw1(sip)---(sip)ipipgw2--tgw(sip)

**Conditions** : When slow start, delay-media to delay-media being used

**Workaround** : There is no workaround

CSCek33537: Codec negotiation failed when SIP EP had mix-codec configured

**Symptom** : SIP EP with different codecs may not able to establish a stable call among them.

**Conditions** : SIP EP on Cisco Unified Communications Manager Express/GW, if different codecs  
(included using voice-class codec configuration) was set under voice register pool.

**Workaround** : There is no workaround.

Further Problem Description: A basic call may be established among the SIP EP. But it was not proper setup with correct codec selection. So the subsequent action on the SIP EP, like transfer/forward/conference, etc. may fail.

CSCsb23025: Locale installer does not include CME spec phrases

**Symptom** : User locale files not available on Cisco Unified Communications Manager Express with 7911 phone.

**Workaround** : Use default locale information.

CSCsb72082: Memory corruption while ending a call

**Symptom** : A Cisco 3845 acting as a SIP gateway crashes when a call is placed from SIP phone to PBX. PBX is an Avaya s8700.

**Conditions** : This occurs on a Cisco 3845 running Cisco IOS Release 12.3(11)T9

**Workaround** : There is no workaround.

CSCsc60509: Memory leak in CCSIP\_UDP\_SOCKET

**Symptom** : Low memory condition caused by a leak in CCSIP\_SPI\_UDP process

**Conditions** : Memory leak will be seen if there is a re-Subscribe for a call context/out of call context subscription. For gateway this typically will be dtmf subscription.

**Workaround** : If possible, do not send re-Subscribe/Un-Subscribe.

CSCsd16947: Spurious memory access traceback seen from 7914 button smashing

**Symptom** : Spurious memory access may be observed on a Cisco Unified Communications Manager Express under certain conditions when multiple DN buttons configured on an IP phone are pressed repeatedly and randomly.

**Conditions** : When multiple DN buttons on a phone are pressed randomly and repeatedly.

**Workaround** : There is no workaround.

CSCsd36869: CME does not put transferred number digits in line focus

**Symptom** : Status line changes to show from xxxx instead of reading transfer xxxx.

**Conditions** : This occurs while receiving a call while in the middle of transferring another call on Cisco Unified Communications Manager Express IP phone.

**Workaround** : There is no workaround.

CSCsd36943: CME sends incorrect language code to phone

**Symptom** : Phone in services mode, displays english text when the user-locale is set to Denmark.

**Conditions** : This occurs when the following output is visible:

```
telephony-service
  user-locale DA
```

**Workaround** : There is no workaround.

Further Problem Description: If the user-locale of the phone is set to Denmark, the phone sends in “dk” as the accept language in the HTTP headers for IP Phone services. The CCM sends in “da” correctly.

CSCsd39342: Call hung when 487 is not received after Cancel

**Symptom** : Call legs are hung and memory leak in the stack

**Conditions** : This issue is seen when a 487 is not received by the SIP gateway after it send out a CANCEL and if the image has the fix for CSCej42804.

**Workaround** : The remote end must send 487 to the CANCEL.

CSCsd46569: CME delayed call-waiting ring/beep

**Symptom** : Call-waiting ring or beep does not give initial tone and is delayed 10 seconds.

**Conditions** : Call-waiting calls on Cisco Unified Communications Manager Express. Initially found in Cisco IOS Release 12.4(2)T2 on both firmware 7.2(2) and 7.2(4).

**Workaround** : There is no workaround.

CSCsd47013: 2nd call callerid not correct with shared lines

**Symptom** : No call information for the second call is displayed on the shared DN for other phones after the first call is terminated.

**Conditions** : Two incomings on the same shared DN. After the first call is terminated, the second call should be presented for the shared DN on all phones.

**Workaround** : There is no workaround.

CSCsd54414: Ringing persists after 1st of 2 shared line calls is answered on another

**Symptom** : The ephone may continue ringing.

**Conditions** : If a call of two incoming calls on a shared DN is answered by an ephone, another ephone having this shared DN may continue ringing.

**Workaround** : Reset the ephone that has this ringing problem.

CSCsd57802: SIP CME: voice register CLI is not available for c2801 adventerprice

**Symptom** : voice register global command is not accepted.

**Conditions** : Under **conf t**, voice register command is not an option.

**Workaround** : There is no workaround.

CSCsd60182: Wrong DN may be chosen for SetCallState on a shared line

**Symptom** : For the shared DN, the wrong DN may be chosen for the sccp msg of SetCallState.

**Conditions** : This problem can't be replicated easily. However, it may be observed after several calls to both channels on a shared DN. When the problem occurs, the ephone would not be able to answer a call and may keep ringing. The debug trace would indicate that a DN not configured on the ephone is chosen when sending the call state sccp message to a line.

**Workaround** : There is no workaround.

CSCsd67460: 7970/71 has erroneous acct, login and flash softkeys

**Symptom** : On 7970s and 7971s running on Cisco Unified Communications Manager Express 4.0, the “Acct”, “Login”, and “Flash” softkeys are wrongly labeled as “No Park”, “Service is”, “CallPark”.

**Conditions** : Any 7970/7971 firmware.

**Workaround** : There is no workaround.

CSCek26630: Real call\_id to be provided in COT REQ to TSP

**Symptom** : COT fails for outgoing calls.

**Conditions** : For the platforms which use TSP for COT processing

**Workaround** : There is no workaround.

Further Problem Description: Changes made to the platform resulted in this requirement. Unless the real call-id is passed, TSP will not be able to complete the COT and call for outgoing calls. This is because TSP must allocate and use the same DSP for COT and the call for outgoing calls. See also CSCek34759.

## Open Caveats - Cisco IOS Release 12.4(4)XC1

There are no open caveats in this release.

## Resolved Caveats - Release 12.4(4)XC

CSCei33312: Fast dial on ephone-dn does not work when ephone trunked to fxo port

**Symptom** : Outbound calls from ip phone to PSTN using personal speed dial fails when the ip phone is configured to be trunked to an fxo port. It appears that digits are sent before the fxo port is seized completely. Inbound calls from PSTN through fxo port onto ip phone work fine.

**Conditions** : Condition only occurs when personal speed-dial is used.

**Workaround** : Do not use trunk configuration

CSCek13657:%SYS-2-NULLCHUNK:Traceback seen at bootup with no startupconfig

**Symptom** : Traceback Conditions: Bootup

**Workaround** : There is no workaround.

CSCek25124: Auto-line x answer-incoming enhancement for customer

CSCek25882: Incorrect KS\_NO\_HLOG4 mask for seized call state

CSCek27807: feature\_Vsa not shown in syslog mode

**Symptom** : The accounting through syslog using gw-accounting syslog does not generate feature-vsa in the output. They are seen on console but not on syslog server.

**Conditions** : The records are generated for each and every call. The problem is reproduced by making a simple A-B call and collecting the output at syslog server. The gw-accounting syslog needed to be configured globally.

**Workaround** : There is no workaround.

Further Problem Description: The problem is a Cisco IOS software bug. The generated log did not use the errmsg. Only error messages are sent to syslog.

CSCei77396: enable ip routing causes error msg %FIB-2-IF\_NUMBER\_ILLEGAL: Attempt

**Symptom** : FIB-2-IF\_NUMBER\_ILLEGAL: Attempt to create CEF interface for Loopback0 with illegal if\_number: 0

**Conditions** : This has been consistently encountered in the case of SSO mode of HA, when the active router which does the switchover comes up again as the standby node.

**Workaround** : There is no workaround.

CSCek31887: QSIG rose\_decode\_facilityIE problem

CSCsc42926: One way Video in Skinny-H323 Conferencing

**Symptom** : Conferencing with Cisco Unified Communications Manager Express-Video capable phones results in one-way video, conference initiator's local video screen disappears while remote video screen is still active. The conference participant's local video screen is active while the remote video screen is frozen.

**Workaround** : There is no workaround.

CSCsc45472: Cleanup of dnagling ccb when back to IDLE

CSCsc46027: 1-way voice for call xfer with xfer party in diff gw

CSCsc48792: spurious access - traceback seen at strncpy --> Active\_Facility-->AFW

CSCsc48885: spurious access and tracebacks at ConsultRespComplete

CSCsc60999: No Ringback after call is forwarded with CFNA

**Symptom** : Call originator no longer hear ringback tone after call forward no answer. B has Call Forward No Answer (CFNA) configured. All ports under 12.4(1), 12.4(2), 12.4(3) or 12.4(4) Cisco Unified Communications Manager Express control. 1) A calls B. A gets ringback tone 2) B does not answer. After CFNA timeout, call is forward to C 3) C gets ringing tone. But A on longer gets ringback tone.

**Conditions** : Root Cause seems to be that there is a call flow change on Cisco Unified Communications Manager Express side starting 12.4(1): GW receives a stoptone(ringback) after CFNA timeout followed by starttone(ringback) once call is forwarded to C. When GW side gets start tone(ringback tone), it sends ccCallAlert to ccapi->vtsp. But ccAlert is ignored by VTSP due to voice leg is already in alerting state. Thus call originator no longer hears ringback tone.

CSCsc80794: CPU utilization on 2800 shoots to maximum level after loading the image

**Symptom** : 100% CPU utilization will be observed on Cisco 2811, Cisco 2821, and Cisco 2851 routers even with no or minimal traffic.

**Conditions** : This will happen on the Cisco 2811, Cisco 2821, and Cisco 2851 routers with the images that have integrated the CSCsc10961 fix and have

**Workaround** : Use Cisco Unified Communications Manager Express version XL1 or XL3 which are special Cisco Unified Communications Manager Express release version. Send ccGenerateTone(ringback) rather than ccCallAlert when call is in proceeding state and a startTone(ringback) is received from Cisco Unified Communications Manager Express/SRST.

CSCsc84018:**bulk-speed-dial** config does not appear under **show telephony-service**

**Symptom** : The **bulk-speed-dial** configuration command does not appear under telephony-service. Even though there is **show telephony-service bulk-speed-dial** command, it would be good to show the **bulk-speed-dial** in **show telephony-service** or at least in **show telephony-service all**.

**Workaround** : There is no workaround.

**Conditions** : This symptom occurs whenever a v-access is cloned from the dialer interface and could be PPPoE, multilink or PPPoA.

**Workaround** : Configure the **interface mtu** command to the required value.

CSCsd12178: B-ACD: sh ephone-hunt 1 statistics start issue

**Symptom** : Cisco Unified Communications Manager Express with B-ACD: When the show ephone-hunt 1 statistics start fri 8 to fri 17 command is entered, the statistics show from Friday 00:00 to 10:00 AM. The display should show from Friday 8:00 AM to 17:00 PM.

**Workaround** : There is no workaround.

CSCsd12361: 7960/70/61- Ringer is not stopped after first incoming call is answered

**Symptom** : The IP phone may continue ringing after answering an incoming call.

**Conditions** : If an ephone is running sccp v5 or onwards and there are more than one incoming calls, the phone continues ringing after answering a call by pressing line button or softkey.

**Workaround** : 1. Running sccp v4 or earlier versions phoneload. 2. Answering a call by lifting the headset or pressing the speakerphone button.

CSCEj31527: Qsig BRI ISDN basic call failure

**Symptom** : Basic Qsig calls via BRI interface fail after new code is collapsed.

**Conditions** : Basic Call Flows that fail IP-Phone --> CME-----BRI QSIG----PBX--> Phone2

**Workaround** : There is no workaround.

## Open Caveats - Release 12.4(4)XC

CSCeh35496: **sh isdn active** does not show voice calls for IP Communications Voice/Fax Network Module (back-to-back)

**Symptom** : The **isdn active** command is not able to display the active/voice call.

**Workaround** : There is no workaround.

CSCEj30662: AudioStream go through CME if Ori EP park the call to 3rd local EP

**Symptom** : A video call has been established between 2 ip phones. If the originating endpoint parks, then pickup the call, the video is dropped and the call becomes an audio only call.

**Workaround** : There is no workaround.

CSCek25773: Router crashed when dialer interface come up

**Symptom** : The Cisco 2811 router crashes when a DDR call is established with certain T1/E1 interfaces.

**Conditions** : The crashed problem is found if Ernst T interface is also equipped to the gateway.

**Workaround** : Remove the Ernst T interface from the gateway.

CSCek26954: Qsig: Calling name not shown in called phone at ALERT and CONNECTED

**Symptom** : When a remote Ephone call is made to an Ephone in PBX Cisco Unified Communications Manager Express, the caller name is not displayed on the PBX phone during the ALERT or CONNECTED states.

**Workaround** : There is no workaround.

CSCek28264: Wrong caller id is wrong when transferring a call out of a trunk dn

**Symptom** : Caller id update is not supported when a call on a trunk dn is hairpin transferred out to another destination. The destination party continues to see the transferrer's party rather than the connected party.

**Workaround** : There is no workaround.

CSCsc28501: Live only MoH disconnected not fallback to tone-on-hold

**Symptom** : If only live MoH is configured in Cisco Unified Communications Manager Express and the music source is disconnected, user hears dead air.

**Conditions** : Only MoH live feed is configured, no MoH from flash.

**Workaround** : Configure MoH from Flash.

CSCsb35485:GUI interface cannot add phone if no auto-reg-ephone

**Symptom** : Cisco Unified Communications Manager Express GUI says "no new phone to add" when trying to add a new phone.

**Conditions** : **no auto-reg-ephone** is configured under telephony-service.

**Workaround** : Remove **no auto-reg-ephone telephony-service** (allow auto ephone registration).

CSCsb67970: CME GUI script error

**Symptom** : The following message is issued when a new phone is added:

A script on this page is causing IE to run slowly. If it continues to run, your computer may become unresponsive. Do you want to continue?

**Conditions** : The problem seems to happen only when there are many Ephones and Ephone-dn in the Cisco Unified Communications Manager Express (230 Ephones and 710 Ephone-dn, for instance). The problem stops when the number of Ephones is reduced to 10. If you click on "No" to "Do you want to continue running the script," the IE freezes for about seven seconds and then you are able to add phones. This seems to happen when Cisco Unified Communications Manager Express has a lot of Ephones and/or ephone-dn in the Cisco Unified Communications Manager Express.

**Workaround** : There is no workaround and no functional impact.

CSCsc01531: Router crash when placing more calls than available queue length

**Symptom** : The router may crash when trying to place more calls in the BACD queue than the configured queue length.

**Conditions** : This symptom has been observed when more calls are placed to BACD queue than the configured queue length.

**Workaround** : Set the codec under dialpeer to g711ulaw.

CSCsc04294: BACD AA script crashes on a call after **no moh** and configuring moh again

**Symptom** : BACD TCL Script may crash and calls drop.

**Conditions** : This happens when the **no moh** and **moh** commands are entered under telephony-service after BACD has been configured and loaded. **& Live MoH** is not configured.

**Workaround** : Avoid using the **no moh** and **moh** commands and configure and use Live MoH instead.

CSCsc28450: **show ephone summary** always says MoH Live feed present

**Symptom** : The **show moh summary** command does not update the MoH live feed status. It always says Live MoH Feed present even when the source is disconnected.

**Workaround** : To determine if the music signal is too low and not present, find the voice port sourcing the music and enter the **show voice call port number** command. Check the input level.

CSCsc56843: CCME display not shown in internationalized language when transferred

**Symptom** : Ephone 3 calls Ephone 2 and is connected. When you press the transfer button on Ephone 3, the word “transfer” shows in Japanese. When you press the first digit “1” the display is changed to “transfer 1” in English.

**Conditions** : User-locale is not set to US, United States.

**Workaround** : There is no workaround.

CSCsc74157: Pings fail with using ISDN switch-type primary-qsig

**Symptom** : Ping fails intermittently with the isdn primary-qsig switchtype but it works fine with the isdn primary-5ess switchtype.

**Conditions** : The test has two earnest t2s back-to-back. The test pings the Cisco 3845 serial interface from the Cisco 3725. When the switch-type is set to primary-5ess it always works. When the switch-type is set to primary-qsig, the request fails right away most of the time. Sometimes the first ping works but fails thereafter.

**Workaround** : There is no workaround.

CSCsc85957: Video is not restored on Hold/Resume with G.729 over H323

**Symptom** : Video is not restored after resume when the codec is G.729 and the other leg is H323 and Music on Hold is configured.

**Workaround** : Choose codec G.711u instead of G.729.

CSCsc94466: Traceback at stcapp\_handle\_call\_disconnect\_event (dangling ccb present)

CSCsd10942: Stuck button light on 7914 after repeated pressing

CSCsd11395: Pressing in-use shared dn button on 7970 shows wrong line

**Symptom** : A 7970 phone shares a line with another device. If the other device causes that line to become in-use, and the 7970 user attempts to select that in-use line, the 7970 will open a new call on other lines instead of doing nothing.

**Conditions** : 7970 and shared lines.

**Workaround** : There is no workaround.

CSCsd16892: Cisco VG224 power denial *click click* heard after call park from analog phone

**Symptom** : Continuous power denial click click is heard under the following condition - analog phone A under Cisco Unified Communications Manager Express control with overlay configured - A calls B and connected - A either blind transfer B to C or park B's call

Root cause: When A is in pure overlay mode, Cisco Unified Communications Manager Express sends onhook to each of the overlay number for A with the same call reference. GW side returns with NewCall for each onhook. This causes confusion on Cisco Unified Communications Manager Express side and gets it into infinite loop due to tries to toggle between new call requests.

**Workaround** : GW side should only response with one new call per call ref. Suppression mechanism is now added in GW sw.

CSCsd19496: Missing TsOnHook for ch 2 of hairpin xfer DN has monitor-port-DN LED red

CSCsd35474: Router crashed during the capf upgrade

**Symptom** : Router crashed when performing the CAPF upgrade

**Workaround** : There is no workaround.

CSCsc35814: Memory leak with QSIG-rose running Qsig call transfer regression post syn

CSCsd08105:7960 speaker phone light does not turn off after pressing hold

**Symptom** : When a call is put on hold on a 7960 Cisco IP phone and the call was on speaker phone, resuming on another phone does not clear the speakerphone light (release 7.x phoneload).

**Workaround** : Depress the speakerphone light to clear it.

CSCsd11990: Dialer does not bind the call in unidirectional ping with specific config

**Symptom** : Two routers are connected back-to-back; pings work in one direction but fail in the other direction. The SETUP is received, but the call does not bind and instead gives a dialer error message.

**Conditions** : In this case, multiple dialer interfaces are configured under same pool. The configuration should have dialer calling numbers to distinguish between the different dialer interfaces in the same pool. In summary, the configuration is not complete to bind the incoming call to a dialer interface.

**Workaround** : You must configure the **dialer-remote name** with PPP authentication enabled.

CSCsd13066: No caller ID displayed for a forwarded call on IP Phone running 7.x

**Symptom** : When release 7.x phoneload is used on a forwarding phone, the forward-to party does not see the forwarded party number on the display.

**Workaround** : There is no workaround.

CSCsd13098: Port monitor ceases to work correctly after re-config of ds0/voice-port

**Symptom** : When removing and reconfiguring ds0-group that is monitored by a trunk dn port monitor command, the existing trunk monitor logic does not recover the new configuration automatically.

**Workaround** : Always reconfigure the trunk dn monitor port command after the port that it monitors is reconfigured.

CSCsd19564: Button optimization for park-recall, pickup-on-hold & pickup-at-alert

**Symptom** : When a trunk dn call that is transferred out of the phone arrives back on a phone that has the shared trunk dn instance, the call is optimized back to that shared trunk dn instance, freeing the transfer-to line button. This operation is done only when the original trunk dn call is transferred. Further work is needed for other operation such as parking the trunk dn call or picking up the call. While the park and pickup operation on a trunk dn call is legal, button optimization is not possible. When the call arrives back on a phone that has the shared trunk dn instance, this call is not optimized back to that trunk dn instance.

**Workaround** : There is no workaround. The call optimization feature is only possible when the trunk dn call is transferred not when it is parked or picked up.

CSCsd25454: Onhook redial of an fxo trunked dn/line may occasionally fail.

**Symptom** : Redial while onhook may not always work for a call that needs to be placed over an fxo trunk dn.

**Workaround** : Offhook redial the call or place the call without redial.

## Caveat Updates and Special Notices

The following information has changed since the initial release of this document:

CSCsd28570

**Symptom** : A vulnerability exists within the Cisco IOS Authentication, Authorization, and Accounting (AAA) command authorization feature, where command authorization checks are not performed on commands executed from the Tool Command Language (TCL) exec shell. This may allow authenticated users to bypass command authorization checks in some configurations resulting in unauthorized privilege escalation.

**Conditions** : Devices that are not running AAA command authorization feature, or do not support TCL functionality are not affected by this vulnerability. This vulnerability is present in all versions of Cisco IOS that support the **tcsh** command.

**Workaround** : This advisory with appropriate workarounds is posted at <http://www.cisco.com/warp/public/707/cisco-response-20060125-aaatcl.shtml>

Further Problem Description: This particular vulnerability only affected Cisco IOS versions 12.3(4)T trains and onwards. (12.3 Mainline is not affected) Please refer to the Advisories “Software Versions and Fixes” table for the first fixed release of Cisco IOS software.

## Additional References

Use these release notes with the documents listed in the following sections:

- [Release-Specific Documents](#)
- [Platform-Specific Documents](#)

## Release-Specific Documents

The following documents are specific to Release 12.4 and apply to Release 12.4(4)XC:

- [\*Cross-Platform Release Notes for Cisco IOS Release 12.4T\*](#)
- [\*Cisco IOS Software Releases 12.4 Special and Early Deployments\*](#)
- [\*Caveats for Cisco IOS Release 12.4\(20\)T\*](#)

## Platform-Specific Documents

Hardware installation guides, configuration and command reference guides, and additional documents specific to the Cisco 2800 series routers are available at:

[http://www.cisco.com/en/US/products/ps5854/tsd\\_products\\_support\\_series\\_home.html](http://www.cisco.com/en/US/products/ps5854/tsd_products_support_series_home.html)

## Cisco IOS Software Documentation Set

The Cisco IOS software documentation set consists of the Cisco IOS configuration guides, Cisco IOS command references, and other supporting documents.

## Documentation Modules

Each module in the Cisco IOS documentation set consists of one or more configuration guides and one or more corresponding command references. Chapters in a configuration guide describe protocols, configuration tasks, and Cisco IOS software functionality, and contain comprehensive configuration examples. Chapters in a command reference provide complete command syntax information. Use each configuration guide with its corresponding command reference. *Cisco IOS Software Documentation* is available in html or pdf form.

Select your release and click the command references, configuration guides, or any other Cisco IOS documentation you need.

## Notices

See the “[Notices](#)” section in *About Cisco IOS Release Notes* located at:

[http://www.cisco.com/en/US/docs/ios/12\\_4/12\\_4x/12\\_4xy15/ReleaseNote.html](http://www.cisco.com/en/US/docs/ios/12_4/12_4x/12_4xy15/ReleaseNote.html)

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Use this document in conjunction with the documents listed in the “[Additional References](#)” section.

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