



RELEASE NOTES

Software 4.0.8 Rev H | April 2016 | 3725-49120-001D

Polycom[®] UC Software 4.0.8 Rev H Release Notes
For SoundPoint[®] IP, SoundStation[®] IP, SoundStation Duo[™],
VVX[®] 500 and VVX[®] 1500 Phones, and SoundStructure VoIP
Interface



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General

Polycom® Unified Communications (UC) software 4.0.8 Rev H is a general release that is distributed to all Polycom partners and customers. These release notes include references to previous UC Software versions to assist administrators who are updating from a previous software release to a UC Software 4.0.x release.

UC Software 4.0.8 Rev H supports SoundPoint® IP, SoundStation® IP, SoundStation Duo™, VVX® 500 and VVX® 1500 Business Media phones, and SoundStructure VoIP Interface. For more details on configuring the phones and the software, refer to the [Polycom® UC Software 4.0.5 Administrators' Guide](#).



Note: Upgrading Updater on SoundStation IP 6000 and SoundStation IP 7000 systems

Polycom strongly recommends upgrading the SoundStation IP 6000 and SoundStation IP 7000 systems to use Updater 5.0.8 to ensure that the Open SSL fix is in place before you install UC Software 4.0.8 Rev H.



Note: Using Polycom Desktop Connector

Polycom strongly recommends using the latest Polycom Desktop Connector (PDC) version 1.0.XXXX.16-MAY-13 17:22, available to download from the [Polycom Desktop Connector web site](#), with UC Software 4.0.8 Rev H.

Important Upgrade Notes and Considerations in UC Software 4.0.8 Rev H

Important Update

Impacted: All Polycom SPIP, SSIP and VVX Business Media Phones.

Details: This release includes a critical fix (VOIP-115900) that corrects a problem with Polycom phones manufactured with Serial Number or MAC address in the range 64167F as opposed to 0004F2. Without this fix, customers using Polycom phones with the new serial number range will see impaired performance.

Recommendation: Polycom recommends that this release be used for all phones going forward. Phones with older Serial Numbers or MAC addresses will continue to work correctly with this build with no impact.

Important Upgrade Notes and Considerations in UC Software 4.0.8 Rev C

This section contains important notes on Polycom hardware and software.

Configuration Files

Since UC Software 3.3.0, Polycom has simplified and extensively modified the configuration files, their respective parameters and defaults, and the provisioning methods. Some of the software updates in UC Software 4.0.8 Rev C may not be compatible with configuration files from software releases prior to UC Software 3.3.0. If you are updating to the UC Software 4.0.8 Rev C from a release prior to 3.3.0, Polycom recommends reading [Technical Bulletin 60519: Simplified Configuration Enhancements in Polycom® UC Software 3.3.0](#) before you deploy the updated software.



Note: Deploying Phones in a Pre-UC Software 3.3.0 Environment

If you use a UC Software version prior to 3.3.0 and want to use a SoundStructure VoIP interface, VVX 500 or SoundStation Duo phone, you need to update your existing configuration files as described in [Technical Bulletin 60519](#).

Upgrading to or Downgrading from UC Software 4.0.x

As of UC Software 4.0.x, Polycom has changed the process of upgrading and downgrading software on the phones. To upgrade to UC Software 4.0.x, you need to install the new Upgrader application. Once you have installed the Upgrader, you can install UC Software 4.0.x.

To downgrade from UC Software 4.0.x, you need first to install a new application called the Downgrader. Once you have installed the Downgrader, you can install an earlier UC Software version.

For detailed instructions on how to upgrade and downgrade your UC Software, see [Engineering Advisory 64731: Polycom® UC Software 4.0.x: Upgrade and Downgrade Methods](#).

Introducing the Integrated Updater Application

In UC Software 4.0.x, the Updater files and the UC Software files are combined in a single, integrated software package. The Updater application is new for the UC Software and replaces the BootROM application incorporated into previous software versions. The name Updater applies to UC Software versions 4.0.0 or later, while the name BootROM continues to be used in UC Software versions earlier than 4.0.0.

Considerations for Legacy Phones

UC Software 4.0.8 Rev C does not include support for the SoundPoint IP 300, 301, 320, 330, 430, 500, 501, 600, 601, SpectraLink 84xx, and SoundStation IP 4000 phones. These phones are termed legacy products and are supported for critical issue fixes in SIP 3.2.x (IP 430), SIP 3.1.x (IP 301, 320, 330, 501, 600, 601, 4000), and SIP 2.1.x (IP 300, 500). For assistance on provisioning methods that support a mix of legacy and current products, refer to [Technical Bulletin 35311: Maintaining Older Polycom® Phones Beyond Their Last Supported Software Release](#).

Supporting Earlier Software versions

For details on platform support for previous software versions, refer to [UC Software/SIP Software Release Matrix](#).

Managing SoundStation IP 7000 Phones with HDX Integration

If your phone deployment includes SoundStation IP 7000 phones with HDX 4000/6000/7000/8000/9000 integration, use the below table [Recommended Software Versions for the SoundStation IP 7000 with HDX Systems](#) as a guide to select a recommended software version.

Recommended Software Versions for the SoundStation IP 7000 with HDX Systems

<i>SoundStation IP 7000 Software Version</i>	<i>Polycom HDX Software Version</i>
3.2.1 or 3.2.2 and BootROM 4.2.0	2.5.0.7, 2.5.0.8
3.2.3 and BootROM 4.2.2	2.6.0, 2.6.0.2, 2.6.1, 2.6.1.3
3.3.1 and BootROM 4.3.1	2.6.1.3, 3.0, 3.0.0.1, 3.0.0.2, 3.0.1, 3.0.2, 3.0.2.1, 3.0.3, 3.0.3.1
4.0.1 and Updater 5.0.1	3.0.3, 3.0.3.1
4.0.2 Rev B and Updater 5.0.1	3.0.4, 3.0.5, 3.1.0, 3.1.0.1
4.0.3 Rev F and Updater 5.0.1	3.1.2, 3.1.1.2, 3.1.1.3
4.0.4 and Updater 5.0.3	3.1.3
4.0.7 and Updater 5.0.4	3.1.6*



Note: Polycom Supports HDX Version 3.1.2 and above

Support is available for HDX Version 3.1.2 and above only. Customers are encouraged to upgrade to HDX Version 3.1.2 and above and the corresponding recommended UC Software suitable for the SoundStation IP 7000.

Managing SoundStation IP 7000 Phones with RealPresence® Group Series Integration

If your phone deployment includes SoundStation IP 7000 phones with RealPresence® Group 300/500/700 integration, use [Recommended Software Versions for the SoundStation IP 7000 with RealPresence Group Series](#) table as a guide to select a recommended software version.

Recommended Software Versions for the SoundStation IP 7000 with Real Presence Group Series

<i>SoundStation IP 7000 Software Version</i>	<i>Polycom RealPresence Group Software Version</i>
4.0.2 Rev B and Updater 5.0.1	4.0.0, 4.0.0.1
4.0.3 Rev F and Updater 5.0.1	4.0.1, 4.0.2
4.0.3F and Updater 5.0.1	4.0.1, 4.0.2
4.0.4 and Updater 5.0.1	4.1.1, 4.1.1.1

<i>SoundStation IP 7000 Software Version</i>	<i>Polycom RealPresence Group Software Version</i>
4.0.5 and Updater 5.0.1	4.1.3,4.1.4, 4.1.5
4.0.7 and updater 5.0.4	4.2.0

Supporting the PinYin Input Mode (XT9 input)

UC Software 4.0.8 Rev C supports Chinese character input mode known as PinYin. This feature is available on VVX 1500 phones and will require a feature license for activation. Refer to [Feature Profile 76554: Using PinYin to Input Chinese Characters on Polycom® VVX® 1500 Phones](#) to learn more.

Supporting the BroadSoft Hoteling Feature

UC Software 4.0.8 Rev C supports the following Polycom phones for use with the BroadSoft Hoteling feature: SoundPoint IP 450, SoundPoint IP 550, SoundPoint IP 560, SoundPoint IP 650, and SoundPoint IP 670.

Supporting BroadSoft Premium Automatic Call Distribution (ACD)

UC Software 4.0.8 Rev C supports the following Polycom phones for use with the BroadSoft Hoteling feature: SoundPoint IP 450, SoundPoint IP 550, SoundPoint IP 560, SoundPoint IP 650, SoundPoint IP 670, and VVX 500.



Web Info: Premium Automatic Call Distribution (ACD)

For detailed information about the ACD feature, see [Feature Profile 76179: Using Premium Automatic Call Distribution for Call Centers](#).

Understanding Phone Platform Features and Licenses

As of release 4.0.0, UC Software supports the Productivity Suite, which includes features such as a Corporate Directory, Visual Conference Management, USB Call Recording, Polycom Desktop Connector, and Exchange Calendar. Upgrading to the current UC Software automatically enables the Productivity Suite; no license is required. The Voice Quality Monitoring (VQMon) feature continues to be a licensed and paid product.

For customers using versions of Polycom UC Software prior to 4.0.0, Polycom provides a site license for all features in the Productivity Suite, except for the VQMon feature. To get a license for the VQMon feature, select *Download a Non VQMon SITE License* on [Polycom Voice Applications](#). You can also find a trial license for the VQMon feature which enables you access to the feature for a limited time.

UC Software 4.0.8 Rev C supports a range of features that are available on the SoundPoint IP phones. You can refer to [SoundPoint IP Features and Licenses](#) table or [SoundStation and VVX Series Features and Licenses](#) table to find out whether, in UC Software 4.0.8 Rev C, a phone does not support a feature (No), a phone supports a feature without a license (Yes), or the phone requires a Feature License to support a feature.

SoundPoint IP Features and Licenses

<i>Feature</i>	<i>IP 321/331/335</i>	<i>IP 450/550/560</i>	<i>IP 650/670</i>
VQMon	Feature License	Feature License	Feature License
Call Recording	No	No	Yes
Conference Management	No	Yes	Yes
4-way local conference	No	Yes	Yes
Electronic Hookswitch	Yes	Yes	Yes
Enhanced Feature Keys	Yes	Yes	Yes
Customizable UI Background	No	Yes	Yes
Asian Languages	Chinese only	Yes	Yes
Configurable Soft keys	Yes	Yes	Yes
Enhanced BLF	No	Yes	Yes
H.323 Video	No	No	No

SoundStation, VVX Series, and SoundStructure VoIP Interface Features and Licenses

<i>Feature</i>	<i>IP 5000</i>	<i>IP 6000</i>	<i>IP 7000</i>	<i>Duo</i>	<i>VVX 1500</i>	<i>VVX 500</i>	<i>SoundStructure VoIP Interface</i>
VQMon	Feature License	No	No	Feature License	Yes	Yes	No
Local Call Recording	No	No	No	No	Yes	Yes	No
Conference Management	No	No	Yes	Yes	Yes	Yes	No
Electronic Hookswitch	No	No	No	No	Yes	Yes	No
Enhanced Feature Keys	No	No	Yes*	No	Yes	Yes	No
Customizable UI Background	No	No	No	No	Yes	Yes	No
Asian Languages	Yes	Yes	Yes	Yes	Yes	Yes	No

<i>Feature</i>	<i>IP 5000</i>	<i>IP 6000</i>	<i>IP 7000</i>	<i>Duo</i>	<i>VVX 1500</i>	<i>VVX 500</i>	<i>SoundStructure VoIP Interface</i>
Enhanced BLF	No	No	No	No	No	Yes	No
H.323 Video	No	No	No	No	Feature License	No**	No
Pinyin Character Entry	No	No	No	No	Feature License	No	No

*Yes - Supported as of UC Software 3.3.0

** Video is not supported on this release, but is available with an optional VVX camera accessory from UC Software 4.1.3 and later software releases

Downloading the Distribution Files

You can download UC Software 4.0.8 Rev C using either the combined file or the split file, both in ZIP file format. For general use, Polycom recommends using the split file that corresponds to the phone model(s) for your deployment. Refer to [Understanding the Split ZIP File](#) table to match the correct software file to your phone model. If you are provisioning your phones centrally using configuration files, download the corresponding file version and extract the configuration files to the provisioning server, maintaining the folder hierarchy in the ZIP file.

Polycom recommends using the split ZIP file when possible for a shorter upgrade time. However, in some deployments, you need to download both the combined and split files. For example, deployments running a BootROM version prior to BootROM 4.0.0 require the combined file version. If you require both file versions, download both ZIP files, extract all the configuration files from the split version and extract the **sip.id** file from the combined file version. All files other than the **sip.id** files are duplicated across the two ZIP file versions.

The current build ID for the **sip.id** and resource files listed in [Understanding the Split ZIP File](#) table and [Understanding the Combined ZIP File](#) table is **4.0.8.1608**.

Downloading the Split ZIP File

Use to match your phone model with the correct split distribution file and to understand the configuration files included in the distribution files.

Understanding the Split ZIP File

<i>Distributed Files</i>	<i>File Purpose and Application</i>
2345-12360-001.sip.id	SIP application executables for SoundPoint IP 321
2345-12365-001.sip.id	SIP application executable for SoundPoint IP 331

<i>Distributed Files</i>	<i>File Purpose and Application</i>
2345-12375-001.sip.ld	SIP application executable for SoundPoint IP 335
2345-12450-001.sip.ld	SIP application executable for SoundPoint IP 450
2345-12500-001.sip.ld	SIP application executable for SoundPoint IP 550
2345-12560-001.sip.ld	SIP application executable for SoundPoint IP 560
2345-12600-001.sip.ld	SIP application executable for SoundPoint IP 650
2345-12670-001.sip.ld	SIP application executable for SoundPoint IP 670
3111-30900-001.sip.ld	SIP application executable for SoundStation IP 5000
3111-15600-001.sip.ld	SIP application executable for SoundStation IP 6000
3111-40000-001.sip.ld	SIP application executable for SoundStation IP 7000
3111-19000-001.sip.ld	SIP application executable for SoundStation Duo
3111-44500-001.sip.ld	SIP application executable for VVX 500
2345-17960-001.sip.ld	SIP application executable for VVX 1500
3111-33215-001.sip.ld	SIP application executable for SoundStructure VoIP Interface
sip.ver	Text file detailing the build-identification(s) for the release
000000000000.cfg	Master configuration template file
000000000000-directory~.xml	Local contact directory template file. To apply on a per phone basis, replace the 0s with the MAC address of the phone and remove '~' from the file name.
applications.cfg	Contains configuration parameters for microbrowser and browser applications
features.cfg	Contains configuration parameters for telephony features
H323.cfg	Contains configuration parameters for the H.323 signaling protocol
reg-advanced.cfg	Contains configuration parameters for the line and call registration and advanced phone feature settings
reg-basic.cfg	Contains configuration parameters for the line and call registration and basic phone feature settings
region.cfg	Contains configuration parameters for regional and localization settings such as time and date and language
sip-basic.cfg	Contains configuration parameters for the VoIP server, softswitch registration
sip-interop.cfg	Contains configuration parameters for the VoIP server, softswitch registration, and interoperability configuration

<i>Distributed Files</i>	<i>File Purpose and Application</i>
site.cfg	Contains configuration parameters that are set on a per site basis
video.cfg	Contains configuration parameters for video connectivity
video-integration.cfg	Contains configuration parameters for SoundStation IP 7000 and Polycom HDX integration
SoundPointIP-dictionary.xml	Includes native support for the following languages: Chinese, Traditional (for IP 321, 331, 335, 450, 550, 560, 650, 670; IP 5000, 6000, 7000, Duo) Chinese, Simplified (for IP 321, 331, 335, 450, 550, 560, 650, 670; IP 5000, 6000, 7000, Duo) Danish, Denmark Dutch, Netherlands English, Canada English, United Kingdom English, United States French, France German, Germany Italian, Italy Japanese, Japan (for IP 450, 550, 560, 650, 670; IP 5000, 6000, 7000, Duo) Korean, Korea (for IP 450, 550, 560, 650, 670; IP 5000, 6000, 7000, Duo) Norwegian, Norway Polish, Poland Portuguese, Portugal Russian, Russia Slovenian, Slovenia Spanish, Spain Swedish, Sweden
SoundPointIPWelcome.wav	Start-up welcome sound effect
LoudRing.wav	Loud ringer sound effect
Warble.wav	Loud ringer sound effect

Downloading the Combined ZIP File

Use [Understanding the Combined ZIP File](#) as a reference guide to each of the files distributed in the combined ZIP file and a brief description of each file. The combined file is required for any phones running BootROM prior to BootROM version 4.0.0: for example, BootROM 3.2.3 Rev B.

Understanding the Combined ZIP File

<i>Distributed Files</i>	<i>File Purpose and Application</i>
sip.ld	Concatenated SIP application executable
sip.ver	Text file detailing build-identification(s) for the release
000000000000.cfg	Master configuration template file
000000000000-directory~.xml	Local contact directory template file. To apply on a per phone basis, replace the 0s with the MAC address of the phone and remove '~' from the file name.
applications.cfg	Contains configuration parameters for microbrowser and browser applications
features.cfg	Contains configuration parameters for telephony features
H323.cfg	Contains configuration parameters for the H.323 signaling protocol
reg-advanced.cfg	Contains configuration parameters for line and call registration and advanced phone feature settings
reg-basic.cfg	Contains configuration parameters for line and call registration and basic phone settings
region.cfg	Contains configuration parameters for regional and localization settings such as time and date and language
sip-basic.cfg	Contains configuration parameters for the VoIP server and softswitch registration
sip-interop.cfg	Contains configuration parameters for the VoIP server, softswitch registration, and interoperability configuration
site.cfg	Contains configuration parameters that are set on a per site basis
video.cfg	Contains configuration parameters for video connectivity
video-integration.cfg	Contains configuration parameters for SoundStation IP 7000 and Polycom HDX integration

<i>Distributed Files</i>	<i>File Purpose and Application</i>
SoundPointIP-dictionary.xml	<p>Includes native support for the following languages:</p> <p>Chinese, Traditional (for IP 321, 331, 335, 450, 550, 560, 650, 670; IP 5000, 6000, 7000, Duo)</p> <p>Chinese, Simplified (for IP 321, 331, 335, 450, 550, 560, 650, 670; IP 5000, 6000, 7000, Duo)</p> <p>Danish, Denmark</p> <p>Dutch, Netherlands</p> <p>English, Canada</p> <p>English, United Kingdom</p> <p>English, United States</p> <p>French, France</p> <p>German, Germany</p> <p>Italian, Italy</p> <p>Japanese, Japan (for IP 450, 550, 560, 650, 670; IP 5000, 6000, 7000, Duo)</p> <p>Korean, Korea (for IP 450, 550, 560, 650, 670; IP 5000, 6000, 7000, Duo)</p> <p>Norwegian, Norway</p> <p>Polish, Poland</p> <p>Portuguese, Portugal</p> <p>Russian, Russia</p> <p>Slovenian, Slovenia</p> <p>Spanish, Spain</p> <p>Swedish, Sweden</p>
SoundPointIPWelcome.wav	Startup welcome sound effect
LoudRing.wav	Loud ringer sound effect
Warble.wav	Loud ringer sound effect

Release History

This following table shows the release history of Polycom® Unified Communications (UC) software 4.0.8 Rev C.

Release History

<i>Release</i>	<i>Release Date</i>	<i>Features</i>
4.0.8 Rev H	April 2016	This release has important field fixes.
4.0.8 Rev C	February 2015	This release has important field fixes.
4.0.8	January 2015	This release has important field fixes.
4.0.7	July 2014	This release has important field fixes.
4.0.6	March 2014	Added support for the improved touchscreen driver for the VVX 500 phone.
4.0.5	December 2013	Added support for GENBAND directory, Global Address Book (GAB) and Personal Address Book (PAB). Added support for Multiple Appearance Directory Number-Single Call Appearance (MADN-SCA) for Shared Line Appearance scenarios.
4.0.4 Rev C	June 2013	Added corporate directory under the Directories soft key in the call transfer menu.

New Features in 4.0.8 Rev C and 4.0.8

This section lists additions, changes, removals, enhancements, and configuration file parameter changes to UC Software 4.0.8 Rev C and UC Software 4.0.8 beside their respective Polycom tracking identification numbers.

Resolved Issues for UC Software 4.0.8 Rev C

The following table lists issues corrected in version 4.0.8 Rev C.

<i>Category</i>	<i>Issue ID</i>	<i>Found in Release</i>	<i>Description</i>
BLF	VOIP-98339	4.0.8 Rev C	BLF line labels configured using <code>attendant.resourceList.x.label</code> now display the correct value and no longer display the SIP URI of the monitored line.

New or Enhanced Features

The version 4.0.8 software version includes the features and functionality of 4.0.7, with the following addition.

- You can now configure Exchange Calendar alert tones on Polycom VVX phones. (VOIP-96381)
Administrators can customize Exchange Calendar alert tones using configuration files only. The parameter `exchange.meeting.reminderType` is an existing parameter that has been updated for the new customization functionality. (Related to VOIP-96381)
- You can now configure the Message Waiting Indicator (MWI) alert tone. (VOIP-94741)
Administrators can customize MWI tones using configuration files only. The parameter `mwi.reminder.enable` is a new parameter that has been added.
- You can now configure whether applications can use the SSLv3 or SSLv2 protocols. (VOIP-97280)
Administrators can customize use of these protocols using the parameter `device.sec.TLS.SSLv2v3.enabled`, a new parameter added with this release.

Configuration File Enhancements

Refer to the following table for a list of all enhancements made to UC Software 4.0.8 configuration file parameters.

<i>Parameter</i>	<i>Permitted Values</i>	<i>Default</i>
call.advancedMissedCalls.addToMissedList	0 or 1	0
<p>If 1, remotely handled calls are logged as missed calls.</p> <ul style="list-style-type: none"> The parameter <code>call.advancedMissedCalls.enabled</code> must also be set to 1. The parameter <code>call.advancedMissedCalls.addToRecievedList</code> is ignored. <p>If 0, remotely handled calls are not logged as missed calls.</p>		
call.doNotPlayLocalOnProvResponseSdp	0 or 1	0
<p>If 1, the local ringback tone does not play after 183 with SDP. If 0, the local ringback tone plays immediately after 183 with or without SDP.</p>		
device.sec.TLS.SSLv2v3.enabled	0 or 1	0
<p>If 1, applications can use SSLv3 (or SSLv2) protocol.</p> <ul style="list-style-type: none"> Note that some applications disable both SSLv2 and SSLv3, regardless of this setting. <p>If 0, applications cannot use SSLv3 (or SSLv2) protocol.</p>		
exchange.meeting.reminderType	0 - 2	2
<p>If 2, all calendar reminders are audible and visual. If 1, the first reminder is audible, followed by silent visual reminders If 0, all reminders are silent visual reminders</p>		
mwi.reminder.enable	0 or 1	0
<p>If 1, a Message Waiting Indicator (MWI) tone plays each time the phone receives an MWI update. If 0, no MWI tone is played.</p>		
tcplpApp.keepalive.tcp.sip.persistentConnection.enable	0 or 1	0
<p>If 0, the TCP Socket connection is closed after 1 minute. When the phone sends a new SIP message, a new connection is opened. If 1, the TCP Socket connection remains open indefinitely.</p>		
volpProt.offerFullCodecListUponResume	0 or 1	1
<p>If 1, the phone offers the full codec list in the resume offer. If 0, the phone does not send the full codec list in the resume offer.</p>		

Resolved Issues

The following table lists issues corrected in version 4.0.8.

<i>Category</i>	<i>Issue ID</i>	<i>Found in Release</i>	<i>Description</i>
Audio	VOIP-95486	4.1.6	New parameter <code>call.doNotPlayLocalOnProvResponseSdp</code> was added to resolve a problem that caused an incorrect local ringback with 183.
Audio	VOIP-94741	4.0.7	New parameter <code>mwi.reminder.enable</code> was added to resolve a problem that caused the phone to play a new message tone when the message count is decreased.
Authentication	VOIP-97199	5.3.0	Previously, telnet, engineering, and technical support modes did not require a valid password after the phone was idle. This problem has been resolved.
BLF	VOIP-96300	4.0.7	A problem was resolved that prevented the BLF line icon and LED from updating correctly when the BLF contacts were removed or added to the pool of monitored lines.
BLF	VOIP-95124	5.2.0	The BLFs assigned to the phone now display correctly on the active line list.
Calling	VOIP-93759	4.1.7	The phone now correctly answers calls from a monitored user.
Calling	VOIP-93928	4.1.6	A problem was resolved that caused a call to be transferred even if the call were hung up before the call was complete.
Calling	VOIP-94452	4.0.5	Calls are now parked successfully on a shared line when you press the Park soft key and enter the extension.
Calling	VOIP-95401	4.0.5	New parameter <code>voIpProt.offerFullCodecListUponResume</code> was added to resolve a problem that caused the phone to send the incorrect codec invite message when a call was resumed.
Configuration	VOIP-97061	5.3.0	The phone no longer establishes a second call when configured with one line per key when the parameter <code>call.callsPerLineKey</code> is set to 1.
Configuration	VOIP-94345	5.1.2	Minimum value 1 will be taken even if 0 is configured for the parameter <code>reg.x.calls.PerLineKey</code> .
Configuration	VOIP-95675	4.0.7	New parameter <code>call.doNotPlayLocalOnProvResponseSdp</code> was added to resolve a problem that caused 180 with SDP to fail.
Debug	VOIP-94288	4.0.5	A problem has been resolved that caused the phone to drop logs when logging is enabled while the phone is in the busy state.
Directory	VOIP-95222	4.0.4	Previously, a Directory search did not retain the prior search order after you opened a Contact entry. This problem has been resolved...

<i>Category</i>	<i>Issue ID</i>	<i>Found in Release</i>	<i>Description</i>
Directory	VOIP-96334	4.0.7	A problem was resolved that caused a busy message to display when you searched the Corporate Directory.
Functionality	VOIP-93363	4.0.4	SoundStation IP 7000 no longer reboots when a user with audio only is added to the conference call and then dropped.
Functionality	VOIP-95524	4.0.7	SoundStation IP 7000 is now able to join calls using Call Action Join.
Functionality	VOIP-95896	4.0.7	The higher memory usage on the latest builds is now resolved.
Hardware	VOIP-95007	4.0.5	The phone no longer disconnects or has an audio issue when an incoming call is answered by using the hook switch button on the headset.
Hardware	VOIP-95499	4.0.7	A SoundStation IP 7000 connected to Group Series was unable to send DTMF during a conference call. This problem has been resolved.
Hardware	VOIP-94727	3.3.5	The VxWorks phone no longer freezes over time and works appropriately.
Hardware	VOIP-96344	4.0.7	SoundStation IP 670 no longer freezes or reboots when LDAP is enabled.
Hardware	VOIP-95939	4.0.6	SoundPoint IP 670 no longer reboots when connected with three expansion modules.
Hardware	VOIP-93722	4.0.4	SoundPoint IP phones no longer freeze over time. Memory leakage issues are fixed.
Headset	VOIP- 93190	4.0.5	A problem was resolved that caused the phone to dial incorrectly when calling a Plantronics Headset.
Network	VOIP-92685	4.0.5	When the phone is configured with an NTP server with a resync period, the phone no longer reboots or generates core dumps.
Network	VOIP-95109	4.0.6	The phone now correctly responds with a successful data push when these parameters are configured: <pre>httpd.enabled is set to 1 httpd.cfg.enabled is set to 0</pre>
Network	VOIP-93529	4.0.4	Previously, the phone experienced a connection issue with the fallback and failover server when re-register on failover is enabled. This problem has been resolved.
Network	VOIP- 93919	4.0.4	The phone now successfully registers and sends UDP keep-alive messages to the new primary server and no longer sends UDP keep-alive messages to the old server.

<i>Category</i>	<i>Issue ID</i>	<i>Found in Release</i>	<i>Description</i>
Networking	VOIP-95544	4.1.6	The parameter <code>tcpIpApp.keepalive.tcp.sip.persistentConnection.enable</code> was added for the ability to honor inbound TCP keepalive messages. This resolved an issue that caused the phone to drop the TCP connection after 120 seconds.
Networking	VOIP-92856	4.0.3	A problem was resolved that caused the phone to stream SIP packets to the PC port.
Provisioning	VOIP-96116	4.0.7	The phone upgrades to UC Software 4.0.8 with no requirement to power cycle when provisioned externally via HTTPS.
Registration	VOIP-93337	5.0.2	A problem was resolved that prevented phones from re-registering correctly after an outage on a network with multiple call servers.
Security	VOIP-93924	4.1.7	A problem was resolved with the certificates included with the system.
Security	VOIP-95249	4.0.7	The Verisign universal root authority certificate is now available in the trusted certificate list.
User Interface	VOIP-97195	4.0.3	The web interface now correctly displays Expansion Module sections with correct spelling.
User Interface	VOIP-97217	4.0.7	A problem has been resolved that kept the user interface from updating correctly after a call was parked or retrieved.
User Interface	VOIP-94874	5.1.0	In the web interface, the configured soft keys now appear appropriately on the active soft key list.
User Interface	VOIP-96281	4.0.7	A problem was resolved that caused the More soft key to display on the BLF incoming call window.
User Interface	VOIP-97090	4.0.7	A problem was fixed that caused the phone to continue showing caller information after the server disabled the phone.
User Interface	VOIP-97033	4.0.7	The phone no longer displays a Service Unavailable message when you try to park a call.
User Interface	VOIP-96466	4.0.7	The phone no longer displays the Start Keypad Logging option in the Advanced Settings menu.
User Interface	VOIP-95758	4.0.7	A problem was resolved that caused the PC Port Mode option to display on the Ethernet menu in the web interface.
User Interface	VOIP-95671	4.0.7	The phone interface and web interface now display the complete certificates list appropriately under the TLS Security menu.

<i>Category</i>	<i>Issue ID</i>	<i>Found in Release</i>	<i>Description</i>
User Interface	VOIP-93205	4.0.4	A problem was resolved that caused a user interface response issue on the SoundStation IP when the system rebooted without a network connection.
User Interface	VOIP-93558	4.0.6	A problem was resolved that prevented the TLS option from displaying in the configuration menu.
User Interface	VOIP-95155	4.0.4	A problem was fixed that prevented users from logging in successfully using the phone menu and prevented returning to idle screen after login.
User Interface	VOIP-96466	4.0.7	The phone no longer displays Start Keypad Logging in Advanced Settings menu.
Video	VOIP-95828	4.0.4	A problem has been resolved that previously caused a one-way video issue when a video call is connected through BroadSoft UC-One client via a Telstra VPN network work.

Documentation Corrections

This section includes updates to correct errors inadvertently included in the *UC Software Administrator Guide 4.0.5*. You can find this guide at support.polycom.com.

Omitted Parameter

The following parameter was omitted from the previous version of the *UC Software Administrator Guide*.

<i>Parameter</i>	<i>Permitted Value</i>	<i>Default</i>	<i>Description</i>
tcpIpApp.snmp.retryDnsPeriod	60 – 86400 seconds	60	Sets a retry period for DNS queries. Note that the DNS retry period you configure is affected by other DNS queries made by the phone. If the phone makes a query for another service such as SIP registration during the configured retry period but receives no response, the Network Time Protocol (NTP) DNS query is omitted to limit the overall number of retry attempts made to the unresponsive server. If no other DNS attempts are made by other services, then the retry period you configure is not affected. If at any time the DNS server becomes responsive to another service, then NTP also immediately retries its DNS query as well.

Incorrect Default Values Listed for <device/> Parameters

The default values of the <device/> parameters vary according to the region to which the phone is shipped. The *UC Software Administrator Guide 4.0.5* incorrectly lists the default values for the <device/> parameters as “null”. Instead, the document should not list default values for the <device/> parameters. (VOIP-95919, VESC-4789)

Known Issues and Suggested Workarounds

The following issues are known to be present in the current release. They will be reviewed for possible fixes in a future release if no reasonable workaround is available.

Known Issues and Suggested Workarounds for UC Software 4.0.8

<i>Category</i>	<i>Issue No.</i>	<i>Found in Release</i>	<i>Description</i>	<i>Workaround</i>
User Interface	VOIP-98086	4.0.8	When you edit a directory entry and save changes, the Save softkey does not clear correctly.	Do one of the following: <ul style="list-style-type: none">• Wait a few minutes and the screen will time out.• Press the left navigation key, and then press the Back key in the top left corner of the screen.
User Interface	VOIP-98141	4.0.8	When you do a quick search in the Genband GAB, the Submit softkey does not display correctly.	Perform the search using Advanced Search.

Known Issues and Suggested Workarounds for Previous UC Software

<i>Category</i>	<i>Issue No</i>	<i>Found in Release</i>	<i>Description</i>	<i>Workaround</i>
User Interface	VOIP-93548		Back soft key is not functioning at times when viewing the details of a contact while navigating through the select button on the dial pad (applies to VVX 1500).	Use view and back soft keys on the phone.
Corporate Directory	VOIP-93433		The phone is not displaying pre-populated text strings [Search (ABC/ASCII)] under search screen of Corporate Directory.	No workaround is currently available.
	26615		Subnet mask forces all packets through gateway when not using DHCP and when using the wrong subnet mask for the network class in use. For example, using 192.168.X.X addresses with a 255.255.0.0 subnet mask. This issue exists in SIP 1.4.x.	Use the correct subnet mask.

<i>Category</i>	<i>Issue No</i>	<i>Found in Release</i>	<i>.Description</i>	<i>Workaround</i>
	26920		Centralized conference fails due to RTP port opening too slowly in some cases.	No workaround is currently available.
	30086		Boot servers running explicit FTPS are not supported.	Use implicit FTPS or HTTPS.
	30371		Pattern generator for tones does not work well in the case of a single repeating chord.	Start the pattern with a short period of silence followed by the desired initial chord. Loop back to the desired initial chord instead of the initial silence.
	33445		LCS Presence and dialing from Buddy Lists does not work across federations.	To dial contacts across federations, program a speed dial with the SIP URI of the contact. There is no workaround for watching Federated Buddy status from the phone.
	37175		If configuration files are used to set the SNTP server address, date validity checking on CA certificates will be ignored for https provisioning.	Set the SNTP server address through the phone UI or use DHCP to inform the phone of the SNTP server address.
	37273		If the custom idle display and idle browser features are both enabled, the phone UI displays incorrectly.	Do not set <code>ind.idleDisplay.enabled=1</code> and enable the Idle Browser at the same time.
	37984		Enabling the Idle bit-map on SoundPoint IP 330 and 320 phones causes the line key labels and dialed digits to be invisible.	Do not use the idle bit-map on 330/320 phones; instead, set <code>ind.idleDisplay.enabled=0</code> .
	41993		Scrolling through the Corporate Directory may not return complete results if results contain Unicode character values > 127 (server does not support sorting).	Start the search in a different location or avoid use of Unicode characters >127 in directories.
	42027		In certain scenarios, the time-stamping in log files of a SoundStation IP 7000 that is used as a secondary/slave device is incorrect.	As of SIP 3.1.0 the occurrence of this issue only relates to the treatment of Daylight savings Time settings.

<i>Category</i>	<i>Issue No</i>	<i>Found in Release</i>	<i>.Description</i>	<i>Workaround</i>
	44764		SRTP processing may cause performance degradation with certain video/audio codec combinations on the VVX 1500.	If SRTP is being used, limit the video bit rate to 384 Kbps.
	46997		Camera brightness adjustment does not work between levels 3 to 6 on the VVX 1500.	No workaround is currently available.
	48905		Jitter parameter is not correctly computed on the SoundStation IP 6000/7000 as per RFC3550.	No workaround is currently available.
	52141		Daisy-chained SoundStation IP 7000 phones sometimes become stuck during software upgrade.	Pressing any key on the phone will continue the upgrade.
	52142		Video connections with CounterPath Eyebeam client on the VVX 1500 do not work if H.263-1998 codec is selected. This was experienced with Eyebeam version 1.5.19.5 build 52345.	Try using a different codec. Try other versions of Eyebeam client as some do work.
	53514		H.264 calls to an HDX9002 device using an MGC 50 Gateway using H.320 result in lip sync issues (applies to VVX 1500).	Set the call for transcoding on the MGC.
	54027		The receiving phone does not re-invite with a new key at the half-life of the key life-time.	Ensure that both ends use the same key life time so that the sending phone will initiate a key re-negotiation.
	54028		Key changes do not function correctly when multiple crypto suites are enabled.	Configure a single crypto suite on the phone.
	54321		The VVX 1500 does not receive video (does receive audio) when calls are initiated from a Tandberg C20 (running 2.0.0.191232) device using SIP.	No workaround is currently available.
	54799		The VVX 1500 transmits H.264 QCIF video to Tandberg MXPs in H.323 calls.	Set the video bit rate on the VVX 1500 to 512 Kbps to avoid the issue.
	54976		H.264 calls to a Tandberg Edge95 MXP device using a Tandberg Gateway using encrypted media (offered but not required) results in distorted audio and no video on the VVX 1500.	Configure system for encryption required.
	54977		H.264 calls to a Tandberg Edge95 MXP device using a Tandberg Gateway result in lip sync issues on the VVX 1500.	No workaround is currently available.

<i>Category</i>	<i>Issue No</i>	<i>Found in Release</i>	<i>Description</i>	<i>Workaround</i>
	59812		Blind transfer to a URL is not successful on the SoundStation IP 7000. Eventually, the URL soft key becomes unavailable.	No workaround is currently available.
	61091		The configuration parameter <code>tcpIpApp.port.rtp.forceSend=1024</code> works only for the SoundStation IP 6000, 7000 and VVX 1500. It does not function properly for SoundPoint IP phones.	No workaround is currently available.
	62387		Adding a new line registration to a phone with BLF causes the notifications (ringing) for the BLF line to display on the previous line. Introduced in UC Software 3.3.1	A phone reset will resolve the issue.
	62482		Server certificate Serial Number is checked against the host name if the outbound proxy is configured.	No workaround is currently available.
	63123		Instead of initiating a new call, an attendant phone plays a reorder tone when a BLF line key is pressed for the second time.	No workaround is currently available.
	63527		Phone sends out INVITE and CANCEL if no provisional response is received.	No workaround is currently available.
	63609		Cannot answer a call using the speaker soft key when DND is enabled and <code>call.rejectBusyOnDnd</code> is set to zero (applies to SpectraLink 84xx).	No workaround is currently available.
	66251		British Telecom Caller ID type is not correctly supported (applies to SoundStation Duo).	No workaround is currently available.
	68815		The phone does not send a <code>CallState=CallConference</code> notification when a conference is established (applies to all SoundPoint IP and SpectraLink 84xx).	No workaround is currently available.
	69502		Confirm Click-to-dial text does not appear on the SoundPoint IP 331 phone when SNTP fails.	Configure SNTP.
	69552		Music on hold (MOH) call dialog does not get terminated when there is an update from the MOH server.	End the call to restore normal state.
	69735		When the phone is registered with a H.323 line, DTMF digits are not sent in the Tel URI call with Ext and Postd options (applies to VVX 500, 1500).	No workaround is currently available.

<i>Category</i>	<i>Issue No</i>	<i>Found in Release</i>	<i>Description</i>	<i>Workaround</i>
	69898		Quick search bar on the SoundPoint IP 321, 330, 331, and 355 accepts only 15 characters when the corporate directory is configured.	No workaround is currently available.
	70480		When the phone is using the Polycom Desktop Connector, the keyboard arrow keys do not support active and inactive call navigation (applies to VVX 500).	No workaround is currently available.
	70728		Software Upgrade does not work if <partnumber>.xml file is not specified as a part of upgrade.custom.server.url configuration value.	Ensure the part-number.xml file is part of the upgrade.custom.serverurl configuration value.
	71386		Soft keys URIs do not function when the phone is in the Enter Number screen (applies to VVX 1500).	No workaround is currently available.
	71800		Users cannot change the user password using the Web Configuration Utility.	Use the phone's user interface to change the user password.
	72082		The phones do not detect a server certificate status change from REVOKED to GOOD until the phone is rebooted (applies to SoundPoint IP 321, 331, 450, 550, 560, 650, and 670, and SoundStation 5000.)	No workaround is currently available.
	72211		An explicitly trusted Intermediate CA fails TLS verification when it is the issuer of a server certificate.	No workaround is currently available.
	72242		Phone is not able to connect to radius server when configured with EAP method as PEAP and inner authentication as GTC (applies to VVX 500).	Recommend to use Cisco ACS server 5.1 or higher.
	72299		When the SoundPoint IP 450, 560, 650 phones are registered with BLA lines, they continue to display remote hold appearances even after the remote BLA resumes the call.	No workaround is currently available.
	72387		After pressing the Transfer soft key, the remote BLA line does not show remote hold status when call.shared.exposeAutoHolds is set to 1.	No workaround is currently available.
	72601		The SoundPoint IP 33x phones fail to dial authorized call when in the Phone Locked state.	No workaround is currently available.

<i>Category</i>	<i>Issue No</i>	<i>Found in Release</i>	<i>Description</i>	<i>Workaround</i>
	72677		When a NOTIFY message with a higher version is sent, the phone re-subscribes to the server and gets a NOTIFY with the correct version, but fails to update the dialog with the state (applies to SoundPoint IP 450/560/650).	No workaround is currently available.
	72898		Hard key external URL mapping requires EFK enabled on the SoundPoint IP 650.	Enable EFK using configuration files.
	73015		The LifeSize Team 220 behaves incorrectly by remaining in a connecting state when there is a call from VVX 1500 over H323.	No workaround is currently available.
	74120		Plantronics Audio 646 DSP USB headset volume control does not work (applies to VVX 500).	Adjust the volume using the volume keys on the phone.
	74533		A phone configured with a Synergy call server displays the incorrect caller ID on the UI for an incoming call (applies to VVX 1500).	No workaround is currently available.
	74650		In an active audio-only call between a PC client and a VVX 1500, the far end video never starts on the PC client when a user presses Add Video.	No workaround is currently available.
	74763		The MKC5 key to upload logs doesn't work (applies to SoundStation Duo).	No workaround is currently available.
	74958		When DND is enabled, the phone misses the call forward message "Fwd:<number>" (applies to VVX 500, 1500, and SpectraLink 84xx).	No workaround is currently available.
	75157		A phone configured with a Synergy call server displays the incorrect soft keys after a "Conference service unavailable" error is shown. This issue exists in UC Software 3.3.3.	No workaround is currently available.
	75195		Hold/Transfer/Conference does not display when the parameter <code>softkey.feature.basicCallManagement.redundant = 0</code> (applies to SoundStation Duo).	No workaround is currently available.
	75229		A phone configured with a Synergy call server displays the local conference UI when establishing a centralized conference using the Join soft key.	No workaround is currently available.

<i>Category</i>	<i>Issue No</i>	<i>Found in Release</i>	<i>Description</i>	<i>Workaround</i>
	75427		The Unified Call Appearance List (UCAL) filtered view times out to the default UCAL view when a user scrolls the filtered list and does not change the focus (applies to VVX 500).	No workaround is currently available.
	75661		The multi key combination shortcuts for uploading logs and rebooting the phone sometimes do not work (applies to VVX 500).	No workaround is currently available.
	75671		When parking a call from the Favorites menu, the call park input dialog (where users enter a park extension) disappears (applies to VVX 500).	No workaround is currently available.
	75759		Numeric data entered using the dial pad on the phone browser cannot be deleted using the dial pad.	Use the virtual keyboard.
	75778		Using Microsoft Lync, if a user dials an invalid extension, the entry is sometimes not logged in the Placed Calls call list.	No workaround is currently available.
	75869		Changing the local contact directory search option from first name to last name, and vice versa, causes the Restart and Save soft keys to be missing on the phone.	Exiting and re-entering the directory fixes the issue.
	75898		Pressing the App hard key on the phone and trying to dial highlighted/focused SIP/Tel URI does not work with the microbrowser (applies to VVX 1500 and 500).	No workaround is currently available.
	76522		In the hoteling call center feature, the phone does not display the status of the call center when there is a special character in the call center name.	The call center administrator can set the names appropriately.
	76655		Using star (*) in the dial string on the SoundStation IP 7000 causes the phone to send it as a dot (.) to the HDX video end points.	Two stars (**) should be used.
	76753		Removing a BLF line from the server causes the speed dial icon to disappear.	Restart or reboot the phone, and the icon will re-appear.
	76881		On a shared call, reorder tone is not played to the user when Resume attempt fails.	No workaround is currently available.

<i>Category</i>	<i>Issue No</i>	<i>Found in Release</i>	<i>.Description</i>	<i>Workaround</i>
	76977		Adding a new registration line changes the BLF-monitored lines label from first/last name to its extension number.	Rebooting the phone will resolve.
	77039		When PTT is enabled, sender name/ID, updated through the parameter reg.x.displayname, does not get updated during the PPT call.	No workaround is currently available.
	77076		When the XT9 input mode is enabled, the phone displays unmatched UIMA-focused items in the first position during XT9 (PinYin) input.	No workaround is currently available.
	77818		In a shared line scenario, the phone does not display the Resume option for a hold call.	No workaround is currently available
	78232		During a remote conference pickup on a shared line, the phone does not display the call appearance and call indicator.	No workaround is currently available.
	78340		Sending several MWI NOTIFY messages within a few seconds of each other may cause the phone to reset.	Avoid sending multiple MWI messages close together.
	78363		In a shared line scenario with calls on hold, the Call Appearance screen times out and does not return the phone to remote active state (applies to SoundStation IP 7000).	Try pressing the Resume or New Call followed by Cancel soft keys.
	81123		During voicemail playback, going off hook and on hook stops the phone from responding to the Menu, New Call, Dir, Dial, Dial pad Number keys and the Volume Increase/Decrease soft keys.	Try pressing the Line1, Hands free, and Headset buttons to make the phone responsive.
	81366		In PSTN mode, the Enter Number screen pops up and obstructs the swapping of calls during a conference call (applies to SoundStation Duo).	Wait for the Enter Number screen to disappear.
	84457		Phone is playing SystemRing.wav ringtone instead of Low Trill Precedence or Ring Splash ringers when the parameter reg.1.ringType is set to ringer13 or ringer14.	Try to change the ringtone using the phone's user interface.
	84700		A loud noise is observed on the phone when the user presses the Conference soft key followed by the Cancel soft key in PSTN mode (applies to SoundStation Duo).	No workaround is currently available.

<i>Category</i>	<i>Issue No</i>	<i>Found in Release</i>	<i>.Description</i>	<i>Workaround</i>
	85142		Observed an incorrect timer update when a media file is paused and then played on the microbrowser.	No workaround is currently available.
	85801		In a PTT scenario, the phone is unable to switch audio termination from speaker to headset by pressing the headset hard key when the page is in progress (applies to SoundPoint IP 335 and IP 670).	No workaround is currently available.
	85908		Phone no longer supports .jpg images (applies to SoundPoint IP 321/331 and IP 335).	No workaround is currently available.
	89407		In a MADN scenario, observed that the received call list on the phone is not getting updated when a MADN call is answered by a different destination.	No workaround is currently available.
	90046		Observed that the paging is not getting initiated by pressing the mute hard key after remapping the Mute hard key for paging (applies to SoundPoint IP 321)	No workaround is currently available.
	90174/905 22		In a GENBAND environment, making video calls from the phone to Group series or HDX is not supported	No workaround is currently available.

Updates to Previous Software Releases

This section lists new, enhanced, and discontinued software features and capabilities in previous software releases.

Understanding updates to UC Software 4.0.7

This section lists additions, changes, removals, enhancements, and configuration file parameter changes to UC Software 4.0.7 beside their respective Polycom tracking identification number.

New or Enhanced Features

There are no new or enhanced features for this release.

Configuration File Enhancements

Refer to the following table for a list of all enhancements made to UC Software 4.0.7 configuration file parameters.

<i>Parameter</i>	<i>Permitted Value</i>	<i>Default</i>	<i>Description</i>
tcplpApp.snmp.retryDnsPeriod	60 – 86400 seconds	60	Set a retry period for DNS queries. Note that the DNS retry period you configure is affected by other DNS queries made by the phone. If the phone makes a query for another service such as SIP registration during the retry period you configure and receives no response, the Network Time Protocol (NTP) DNS query is omitted to limit the overall number of retry attempts made to the unresponsive server. If no other DNS attempts are made by other services, then the retry period you configure is not affected. If at any time the DNS server becomes responsive to another service, then NTP also immediately retries its DNS query as well.

Understanding updates to UC Software 4.0.6

This section lists additions, changes, removals, enhancements, and configuration file parameter changes to UC Software 4.0.6 beside their respective Polycom tracking identification number.

New or Enhanced Features

87581/91102 Added support for the improved touchscreen driver (*applies to VVX 500*).

Configuration File Enhancements

There are no changes to the configuration file parameters.

Understanding Updates to UC Software 4.0.5

This section lists additions, changes, removals, enhancements, and configuration file parameter changes to UC Software 4.0.5 beside their respective Polycom tracking identification number.

New or Enhanced Features

83102 Added support for Multiple Appearance Directory Number-Single Call Appearance (MADN-SCA) for Shared Line Appearance Scenarios.

87263/72669 Added support for Global Address Book (GAB).

87264 Added support for Personal Address Book (PAB).

87029/88933 The Message Waiting Indicator (MWI) LED light no longer blinks when the phone is in power-saving mode and has not received any new messages.

Configuration File Enhancements

Refer to the following table for a list of all enhancements made to UC Software 4.0.5 configuration file parameters.

Software Version 4.0.5 – Configuration File Parameter Enhancements

<i>Parameter</i>	<i>Permitted Value</i>	<i>Default</i>	<i>Description</i>
Feature.corporateDirectory.alt.enabled	0 or 1	0	Enables the GAB functionality.
Sec.TLS.SOPI.strictCertificateValidation	0 or 1	1	Determines certification validation. When set to 0, there is no certification validation needed and any certification is accepted and trusted.
reg.x.bargeInEnabled	Maximum line number registered	0	Determines whether a user can barge in on an active call.
sec.TLS.SOPI.cipherList	String (max of 1024 characters)	NoCipher	Defines the cipher list for SOPI transactions.

<i>Parameter</i>	<i>Permitted Value</i>	<i>Default</i>	<i>Description</i>
dir.local.serverFeatureControl.method	GENBANDSOPI	<Blank>	Sets GENBANDSOPI to synchronize to the server. If this is not set, no synchronization is done and the standard contact directory is used.
dir.local.genband genband1.serverFeatureControl.reg	1 to Maximum line number registered	1	Determines the line that is used to obtain PAB information. Does not change how numbers are dialed. Does not require a reboot –automatically updates at runtime
msg.mwi.x.led	0 or 1	1	Enables and disables the LED Message Waiting Indicator on receiving a new message
mwi.backLight.disable	0 or 1	0	If set to 1, disables backlight on new voice message arrival.
reg.1.server.1.specialInterop	GENBAND	standard	Required to enable MADN-SCA on GENBAND.
sec.TLS.profileSelection.SOPI	PlatformProfile1, PlatformProfile2 ApplicationProfile1 to ApplicationProfile7	PlatformPr ofile1	Defines the certificates to be used for SOPI authentication.

Note: Also refer to configuration parameter changes in UC Software 4.0.4, 4.0.3 Rev F, 4.0.2 Rev B and 4.0.2.

Understanding Updates to UC Software 4.0.4

This section lists additions, changes, removals, enhancements, and configuration file parameter changes to UC Software 4.0.4 beside their respective Polycom tracking identification number.

New or Enhanced Features

- 35157** Added ability to dial '+' character by pressing '*' twice for international dialing.
- 51255** Added SDP negotiation of "a=rtcp-fb" for RTCP Feedback Messages to request I-Frames.
- 72440** Added support for MS 2008 Radius Server.
- 76188** Added ability to process DTMF entry during early media.
- 78976** Added support for early media followed by local ring back.

Configuration File Enhancements

Refer to the following table for a list of all enhancements made to the UC Software 4.0.4 configuration file parameters.

Software Version 4.0.4 – Configuration File Parameter Enhancements

<i>File</i>	<i>Modification</i>	<i>Parameter</i>	<i>Default</i>	<i>Permitted Values</i>
feature	Added	reg.X.teluri	0	0 or 1
feature	Added	volpProt.SIP.supportFor100rel	1	0 or 1
feature	Added	video.dynamicControlMethod	0	0 or 1
feature	Added	video.forceRtcpVideoCodecControl	0	0 or 1
feature	Added	volpProt.SIP.useLocalTargetUri ForLegacyPickup	0	0 or 1
feature	Added	reg.1.useLocalTargetUriFor LegacyPickup	0	0 or 1
feature	Added	volpProt.SIP.fmtmMustForG7221	0	0 or 1
feature	Added	volpProt.SDP.offer.rtcpVideoCodec Control	0	0 or 1
feature	Added	tcpIpApp.sntp.Aquery	0	0 or 1
feature	Added	lcl.datetime.date.dateTop	1*	0 or 1
feature	Added	lcl.ml.lang.clock.X.dateTop	1*	0 or 1
feature	Added	pstn.fxsType	CO	0 to 256
feature	Added	dialplan.digitmap.timeOut	3 3 3 3 3 3 3	0 to 768

Note: Also refer to configuration parameter changes in UC Software 4.0.3 Rev F, 4.0.2 Rev B and 4.0.2.

*This value is '0' by default for the VVX 500.

Understanding Updates to UC Software 4.0.3 Rev B

This section lists additions, changes, removals, enhancements, and configuration file parameter changes to UC Software 4.0.3 Rev B beside their respective Polycom tracking identification number.

**Note: 4.0.x Software Updates**

Polycom is releasing UC Software 4.0.3. Rev B for AT&T only. Please refer to [Updates to Previous Software Releases](#) for all available software features and updates.

New or Enhanced Features

There are no new changes or enhanced features from previous software versions.

Understanding Updates to UC Software 4.0.3 Rev F

This section lists all changes, additions, removals, enhancements, and configuration file parameter changes to UC Software 4.0.3 Rev F beside their respective Polycom tracking identification number.

**Note: Upgrade to UC Software 4.0.3 Rev F**

If you are currently using UC Software 4.0.3 (build ID 4.0.3.7314), upgrade to UC Software 4.0.3 Rev F (build ID 4.0.3.7562).

New or Enhanced Features

There are no new or enhanced features from previous software versions.

Configuration File Enhancements

There are no changes to the configuration file parameters.

Understanding Updates to UC Software 4.0.3

This section lists all changes, additions, removals, enhancements, and configuration file parameter changes to UC Software 4.0.3 beside their respective Polycom tracking identification number.

**Note: UC Software 4.0.3 is a Limited Release**

UC Software 4.0.3 is a limited release that was distributed only to select partners and customers. The build ID for this release was 4.0.3.7314.

New or Enhanced Features

78271 Added support for tel URI in the P-Preferred-Identity header of INVITE messages.

79601/79006/79007 Added support to allow only RFC3264 type SDP media negotiation.

Configuration File Enhancements

Refer to the following table for a list of all enhancements made to the UC Software 4.0.3 configuration file parameters.

Software Version 4.0.3 – Configuration File Parameter Enhancements

<i>File</i>	<i>Modification</i>	<i>Parameter</i>	<i>Modification Description</i>
feature	Added	volpProt.SIP.useRFC3264HoldOnly	1 Phone always uses RFC3264 during hold 0 Phone uses RFC3264 or RFC2543 during hold based on the response from far end (default)

Note: Also refer to configuration parameter changes in UC Software 4.0.2 Rev B and 4.0.2.

Understanding Updates to UC Software 4.0.2 Rev B

This section lists additions and changes, removals, enhancements, and configuration file parameter changes to the UC Software 4.0.2 Rev B beside their respective Polycom tracking identification number.

New or Enhanced Features

72403 Added support for DHCP renews after loss and recovery of Wi-Fi LAN connection (*applies to SpectraLink 8400*).

76730 Enhanced the digitmap by removing the prepending '+' to the outbound calls and giving the option of configuring the "+" in the dial plan (applies to Lync mode only). Note: For more information on digitmap, refer to the Microsoft Lync Interoperability section.

77038 Added support for early media followed by local ring back.

Configuration File Enhancements

Refer to the following table for a list of all enhancements made to the UC Software 4.0.2 Rev B configuration file parameters.

Software Version 4.0.2 Rev B – Configuration File Parameter Enhancements

<i>File</i>	<i>Modification</i>	<i>Parameter</i>	<i>Modification Description</i>
Wireless	Added	device.dhcp.releaseOnLinkRecovery	1 Phone performs a DHCP release on network link recovery (default) 0 Phone does NOT perform a DHCP release on network link

Note: Also refer to configuration parameter changes in UC Software 4.0.2.

Understanding Updates to UC Software 4.0.2

Polycom distributed UC Software 4.0.2 to select partners and customers. It was shipped with SoundStation Duo. The build-ID for this release was UC Software 4.0.2.8017.

New or Enhanced Features

40451/75448 Added support for XT9 PinYin input for Chinese characters (*applies to VVX 1500*).

52485/66494 Added support for BroadSoft Hoteling Event Package.

57167/66494/76023 Added support for BroadSoft Call Center Status Event Package.

54576 Added support for the new SpectraLink 8452 Wi-Fi handset with 2D barcode reader.

Configuration File Enhancements

Refer to the following table for a list of new parameters. Note that these configuration parameters are detailed in [Feature Profile 76179: Hoteling and ACD](#), which will be made available on [Polycom Profiled UC Software Features](#).

Software Version 4.0.2 – Configuration File Parameter Enhancements

<i>File</i>	<i>Modification</i>	<i>Parameter</i>	<i>Modification Description</i>
feature	Added	feature.callCenterStatus.enabled	Call feature parameter
feature	Added	feature.hoteling.enabled	Call feature parameter
feature	Added	hoteling.reg	Call feature parameters
wireless	Added	barcode.X.Y	Parameters used to configure the 2D barcode scanner*

*For a detailed description of new parameters specific to the SpectraLink 8400 product family, their properties and values, refer to the Polycom SpectraLink 8400 Series Wireless Telephone Deployment Guide.

Understanding Updates to UC Software 4.0.1B

There are no functional differences between Polycom UC Software 4.0.1B and Polycom UC Software 4.0.1. Polycom UC Software 4.0.1B was released to include the VVX 500 and SoundStructure VoIP Interface sip.ld files in the software release package.

Understanding Updates to UC Software 4.0.1

This section lists additions and changes, removals, enhancements, and configuration file parameter changes to UC Software 4.0.1 beside their respective Polycom tracking identification number.

New or Enhanced Features

- 48734** In a server-based, centralized conference, the phone can now send parallel REFERs without waiting for a 202 Accepted.
- 67081/70634** Added support for phones to interoperate with a limited set of Microsoft® Lync™ server features (*applies to SoundPoint IP 321, 331, 335, 450, 550, 560, 650, 670, VVX 500, 1500, SoundStation IP 5000, SoundStation Duo, and SpectraLink 84xx*).
- 67090** Syslog now includes the ability to identify multiple audio streams (*applies to SpectraLink 84xx*).
- 67594** Added interoperability between the Message Waiting Indicator (MWI) and Microsoft Lync.
- 68500** The SpectraLink 84xx handsets now display the X-Loader version information in the Phone menu (Menu > Settings > Status > Platform > Phone).
- 68602** Added support for SS RTP.
- 68798** Added support for Microsoft SRTP extensions.
- 69962/70924** Added Microsoft OCS/Lync Presence functionality to phones (*applies to SoundPoint IP 321, 331, 335, 450, 550, 560, 650, 670, VVX 500, 1500, SoundStation IP 5000, SoundStation Duo, and SpectraLink 84xx*).
- 70122** Phones now display the Away presence status after a period of user inactivity specified by the following parameters: `pres.idleTimeout.offHours.period`, `pres.idleTimeout.officeHours.period`, `pres.idleTimeout.offHours.enabled`, and `pres.idleTimeout.officeHours.enabled`.
- 70232** Added a parameter `call.transfer.blindPreferred` to control whether the Transfer soft key on the SpectraLink 84xx should be a consultative transfer or blind transfer.
- 70614** Added support for Microsoft 2008 Radius (802.1X).
- 71025** Added new per-registration configuration options for several SRTP parameters: `reg.x.srtp.enable`, `reg.x.srtp.offer`, `reg.x.srtp.require`.
- 71183** Added missing barcode symbologies (*applies to SpectraLink 84xx*).

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- 71198** Added an option in the Web Configuration Utility for SIP and Provisioning TLS applications to make the Common Name of Subject test configurable.
- 71424** Updated the presence icon on the phones to be consistent with the Microsoft Lync/OCS style (*applies to SpectraLink 84xx*).
- 71439** Not including the parameter `oai.userID` in the configuration file or setting the value to NULL both result in the phone using its MAC address to check in into the OAI server (*applies to SpectraLink 84xx*).
- 71660** Enhanced the Reset to Default option in the Updater to match the option in the application software.
- 71774** The call forward status on the status bar now displays when Forward – No Answer or Forward – Busy is enabled (*applies to SpectraLink 84xx*).
- 71997** Added full support for RFC2782 (DNS load balancing).
- 72074** In the Web Configuration Utility, the Country Code field has been renamed to Regulatory Domain (*applies to SpectraLink 84xx*).
- 72193** In the phone menu, the Country Code field has been renamed to Regulatory Domain (*applies to SpectraLink 84xx*).
- 72279** Enabled an EFK to allow a user to invoke the Call Back feature while on hook (*applies to VVX 1500*).
- 72304** The default value for the configuration parameter `up.useDirectoryNames` is now 1 (enabled).
- 72310/74129** The SpectraLink series handsets can now display the BootL1 version information in Phone menu (Menu > Status > Platform > Phone).
- 72319** The phone displays a warning icon when the WLAN Network Manager detects an invalid Regulatory Domain request (*applies to SpectraLink 84xx*).
- 72320** The phone displays a warning triangle when the WLAN Network Manager detects an invalid Regulatory Domain limit setting (*applies to SpectraLink 84xx*).
- 72367** The phone automatically publishes an Inactive (Idle) presence status after 5 minutes of user inactivity.
- 72554** Added the ability to configure the `pres.idleTimeout` parameters through the phone menus.
- 72555** Added the ability to configure the `pres.idleTimeout` parameters through the Web Configuration Utility.
- 72654** The Exchange Calendaring feature on the SpectraLink handsets has been improved with the following enhancements:
- The Calendar icon is shown in the main menu once the calendar is authorized.
 - The phone displays a Calendar: synchronizing scrolling message in the status bar.
- 72791** The microbrowser on the SoundPoint phones has been functionally improved with the following enhancements:
- The audio tag element will inject a Play soft key when in focus.
 - The user can now specify additional attributes to the audio tag which will be interpreted as a soft key, thus allowing the user to do things such as a Details soft key.

- The audio tag will have a descriptive label which will be used as the button label for the audio element in the page. This enables each audio tag to be rendered as a single element in the page with an icon and a descriptive text. The user no longer needs to switch to the text to see the details.
- The descriptive label will also be used for the title of the playback screen.

72823 In the media player, the Exit soft key has been renamed to Back.

72824 Playback automatically starts when selecting an audio element from the browser.

73420 When the phone language is set to Japanese, the phone now uses the English AM/PM string for the time/date display (*applies to all SoundPoint IP, all SoundStation IP, and SoundStation Duo*).

73500 The 'Connect/disconnect from the server' option has been moved to the Calendar menu (Features > Calendar) in the SpectraLink 84xx handsets.

73510 The Web Configuration Utility language now supports multiple default language labels and help text in English, with the option to add/access other languages.

73669 Updated the 2048-bit Trusted CA Root Certificate List from VeriSign.

73670 Added new VeriSign Intermediate CA certificates.

73671 Added RSA 2048 V3 Root Certificate to Root Store to all phones.

73805 The phone can now display up to 4 Chinese characters in the soft keys (*applies to SoundPoint IP 450*).

73907/74289 Added the ability to automatically upgrade the BootL1 and BootBlock (*applies to SpectraLink 84xx*).

74247 In the Web Configuration Utility, the default available utility languages depend on the platform.

74417 The Updater (BootROM) now supports Basic Authentication with HTTP/HTTPS.

75308 Added the ability to upload encrypted call lists to the provisioning server (*applies to SpectraLink 84xx*).

75469 The volume of PTT audio has been increased and setting the parameter `voice.handsfree.rxag.SL8440=10`, then updating the phone using Update Configuration does not cause the phone to restart (*applies to SpectraLink 84xx*).

Configuration File Enhancements

Refer to [Software Version 4.0.1 – Configuration File Parameter Enhancements](#) table for a list of enhancements made to the UC Software 4.0.1 configuration file parameters.



Web Info: Parameters Changed in UC Software 4.0.1

The following table includes parameters modified in UC Software 4.0.1. You can find detailed descriptions of the parameters and their values in the [Polycom UC Software 4.0.1 Administrators' Guide](#).

Software Version 4.0.1 – Configuration File Parameter Enhancements

<i>File</i>	<i>Modification</i>	<i>Parameter</i>	<i>Modification Description</i>
sip-interop	Added	call.transfer.blindPreferred	Call feature parameter
debug	Added	feature.lyncDebug	Call feature parameter
site	Added	reg.x.srtp.enable reg.x.srtp.offer reg.x.srtp.require	Call feature parameters
wireless	Added	np.custom1.ringing.toneVolume.usbHeadset	Notification profiles parameter
wireless	Added	np.meeting.ringing.toneVolume.usbHeadset	Notification profiles parameter
wireless	Added	np.normal.ringing.toneVolume.usbHeadset	Notification profiles parameter
wireless	Added	np.silent.ringing.toneVolume.usbHeadset	Notification profiles parameter
site	Added	sec.encryption.upload.callLists	Security parameter
sip-interop	Added	sec.srtp.mki.length	Security parameter
sip-interop	Added	sec.srtp.padRtpToFourByteAlignment	Security parameter
reg-advanced, site	Added	up.headset.phoneVolumeControl	User preferences parameter
debug	Added	up.headset.AlwaysUseIntrinsicRinger	User preferences parameter
reg-advanced, site	Added	up.idleStateView	User preferences parameter
video	Added	video.iFrame.delay	Video parameter
debug	Added	video.iFrame.period	Video parameter
techsupport	Added	voice.usb.headset.rxdg	Audio parameter
techsupport	Added	voice.usb.headset.txdg	Audio parameter
site	Added	voice.volume.persist.usbHeadset	Audio parameter
sip-interop	Added	volpProt.SIP.conference.parallelRefer	Call feature parameter

<i>File</i>	<i>Modification</i>	<i>Parameter</i>	<i>Modification Description</i>
site	Added	webutility.language.plcm ServerUrl	Web Configuration Utility parameter
techsupport	Removed	voice.gain.rx.digital.headset.IP_330	Audio parameter
techsupport	Removed	voice.gain.rx.digital.headset.IP_335	Audio parameter
site	Changed	dns.cache.NAPTR.x.ttl dns.cache.SRV.x.ttl dns.cache.A.x.ttl	The maximum value increased from 65535 to 2147483647
reg-advanced, site	Changed	up.useDirectoryNames	The default value changed from 0 (disabled) to 1 (enabled).
pstn	Changed	up.operMode	The default value changed from 0 to auto.
techsupport	Changed	voice.headset.rxag.adjust. IP_335	The default value changed from -11 to 4 and the maximum value changed from -11 to 90.
techsupport	Changed	voice.headset.rxag.adjust. IP_330	The default value changed from -5 to 4 and the maximum value changed from -5 back to 90.
sip-interop	Changed	VolpProt.SIP.failoverOn503 Response	The default value changed from 1 (enabled) to 0 (disabled).

Understanding Updates to UC Software 4.0.0

This section lists additions and changes, removals, enhancements, and configuration file parameter changes to UC Software 4.0.0 beside their respective Polycom tracking identification number.

New or Enhanced Features

- 26549** Enhanced the local missed call feature for shared line appearances. This feature supports RFC 3326 Reason Header.
- 28514** Enhanced the method of selecting a ring type on the menu screen.
- 29056** Enhanced the method of notifying the user of unregistered lines.
- 30251** Added support for non-volatile call lists (*applies to SpectraLink 84xx*).
- 30887** Added support for 802.1X authentication. Authentication methods include MD-5, EAP-PEAP, EAP-FAST, EAP-TLS, and EAP-TTLS.
- 32169** The user is now notified with a confirmation when deleting contact information.
- 33546** Added a host name field to the DHCP registration.
- 35170** Added support for User Profiles. Users may log into and out of the phone using a server-independent, configuration file-based, authentication method. When successfully authenticated, the user's personal configuration files are applied as well as the user's personal local contact directory and call lists.
- 35171** Updated most configuration parameters to be updated without the need of a reboot. Only a select number of configuration parameters require a reboot in order to be invoked.
- 36166** Added the option for the user to allow ringer volume levels to persist after the phone reboots.
- 38201** The Web-based configuration utility no longer requires the user to submit changes along with a reboot after each page has been modified.
- 41429** Added the ability in the microbrowser to manage allowable characters into the input field.
- 41430** Added the ability in the microbrowser for the user to select an item from a list using the dial pad.
- 44258** Enhanced the API by enhancing the HTTP Push capability by supporting mutual TLS.
- 44699** Added a Reset to Factory capability.
- 45777** Added user accessible diagnostic functions ping and traceroute.
- 47730** The scrolling status bar has been enhanced. The time between scrolled lines has been increased (*applies to SoundPoint IP 3xx*).
- 47766** The Trusted CA Pool Management capability has been enhanced. The number of supported customer certificates has been increased to six.
- 48714** Added ability for the phone to mute the microphone when auto-answering a call.
- 48750** The Web-based configuration utility now enables the user to configure outbound proxies on a per line basis.

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- 48757** Contacts added to the list are now highlighted and displayed without the need to scroll up or down to view the addition.
 - 50258** Enhanced the method of notifying the user of error and warning indications.
 - 51101** Added the ability to use an Emergency Location Identification Number (ELIN) from LLDP-MED to add a P-Asserted-Identity when using emergency routing:
`dialplan.routing.emergency.preferredSource=[ELIN|Config]` (default ELIN)
`dialplan.routing.emergency.outboundIdentity=xxxxxxxx` (default null)
`dialplan.routing.emergency.outboundIdentity.lldp=xxxxxxxx` (default null).
 - 51471** Added a configuration option to disable the test of subject's CommonName against the registration address (associated with CA management).
 - 52844** Added certificate validation for 802.1X.
 - 53128** Added a configuration option to modify the Backlight timeout duration.
 - 53360** Added the ability to display the phone's current ARP table in the diagnostic menu (*applies to SpectraLink 84xx*).
 - 53908** The web-based configuration utility now offers the ability to configure soft keys and line keys.
 - 54301** The timestamp is now displayed alongside the caller in the Call Lists.
 - 54648** Added HTTPS support in the Updater that was previously called BootROM.
 - 54680** Introduced the ability to import and export local and global configuration files using a PC browser.
 - 54683** The browser-based SW Upgrade button that enables user to upgrade phones with one of multiple compatible software versions is available on the Polycom provisioning server.
 - 54730** Noticeable enhancement from the time the phone is powered up and when it is ready for use.
 - 56150** Added Data Link Layer L2 Discovery between phones and PC.
 - 56187** Added ToID and FromID in SIP Publish packets for VQMon reports.
 - 56274** Added multicast group paging based on the SpectraLink PTT solution.
 - 56942** Configuring Soft key (EFK) settings no longer require a reboot in order to take effect.
 - 57392** Added support to the microbrowser for HTTP proxy authentication (*applies to SpectraLink 84xx*).
 - 57981** Added support for custom device certificates.
 - 58007** Added the ability to revoke certificates used in SSL transactions by using OCSP.
 - 58336** Added SHOULD SDP answer behavior as per RFC 3264.
 - 58507** Enhanced the Web-based configuration and provisioning utility.
 - 60297** Added the ability of random distribution of polling to check for software upgrades.
 - 60907** Added the ability to disable Call Waiting while still allowing further outgoing calls.
 - 61051** Added the ability to display custom soft keys on input forms in the microbrowser.
 - 61138** Added support for incoming TLS connections on the Web server.
 - 61343** Added the ability to disable authentication verification for received SRTP packets.
 - 62671** Added a time-stamped log event indicating when the phone is ready to be used.

- 63592** Added API calls to the microbrowser for Media Player.
- 63629** Added Sennheiser EHS configuration menus and options.
- 64144** The alerting LED and associated line-key animation for second and subsequent incoming calls are now disabled when the Call Waiting feature waiting is disabled.
- 64243** The API Data push message size limit has been increased to 2048 bytes from 1024 bytes.
- 64359** Converted the BLA dialog rendering from *No* to *Yes* for user agents that are a remote party to the existing call dialog.
- 65287** Added the ability to prevent a phone from being provisioned at start-up.
Configuration parameter `prov.startupCheck.enabled` [default = 1 (enabled)]
- 66212** Added support for setting the syslog server address from DHCP.
- 66323** Added an administrator operations menu in the Updater to the setup menu: Reboot, Reset Settings, Format File System, and Install BootBlock.
- 66604** The phone reports connectivity event notifications to an 802.1x enabled switch port when a non-authenticated PC disconnects or reconnects to the phone.
- 67600** Password and other security entry fields now perform a brief echo of entered characters before being obscured from view.
- 68110** Added control of available telephony features on the Office Communications Server (OCS) using the `reg.x.telephony` configuration parameter.
- 68899** Added the ability on the microbrowser to enter two-digit dial pad values for selecting entries in a list.
- 69225** Added the ability to allow a hard key to be directly assigned an Enhanced Feature Key (EFK) style macro.
- 69442** Productivity features such as LDAP, Local Call Recording, and Visual Conference Management are enabled without the requirement of a license file. Note that VQMon will remain a licensable feature.
- 71633** The Reset setting in the Updater menu does not erase the CA and Device Certificates.

Configuration File Enhancements

Certain groups of configuration parameters have been modified in UC Software 4.0.0. In these cases, instead of listing every parameter, the following table will specify a group of related parameters with an abbreviated XML path name ending with (.*)

For example, suppose the following parameters are modified: `device.wifi.enabled`, `device.wifi.ipAddress`, and `device.wifi.ssid`. Since these parameters all begin with `device.wifi`, [Software Version 4.0.0 – Configuration File Parameter Enhancements](#) table abbreviates these parameters as `device.wifi.*`

**Note: Parameters with .set**

Most device parameters have identical parameters ending with **.set**. The **.set** parameters are not included in the following table.

**Web Info: Parameters Changed in UC Software 4.0.0**

The following table includes parameters changed in UC Software 4.0.0. You can find the new descriptions and values in the [Polycom UC Software Administrators Guide](#).

Software Version 4.0.0 – Configuration File Parameter Enhancements

<i>Modification</i>	<i>Configuration Parameter</i>	<i>Description</i>
Discontinued	<code>apps.x.Label</code>	Productivity Applications parameter
Discontinued	<code>apps.x.Url</code>	Productivity Applications parameter
Discontinued	<code>apps.ucdesktop.IP</code>	Productivity Applications parameter
Discontinued	<code>apps.ucdesktop.name</code>	Productivity Applications parameter
Discontinued	<code>apps.ucdesktop.port</code>	Productivity Applications parameter
Discontinued	<code>device.auth.*</code> The parameters <code>device.auth.localAdminPassword</code> and <code>device.auth.localUserPassword</code> have not been removed	Provisioning parameters
Discontinued	<code>device.dhcp.offerTimeout</code>	Provisioning parameter
Discontinued	<code>device.prov.appProvString</code>	Provisioning parameter
Discontinued	<code>device.prof.appProvType</code>	Provisioning parameter
Discontinued	<code>device.sec.SSL.*</code>	Provisioning parameters
Discontinued	<code>device.sec.deviceCertEnabled</code>	Provisioning parameter
Discontinued	<code>exchange.server.address</code>	Productivity Applications parameter
Discontinued	<code>httpd.lp.port</code>	Provisioning parameter
Discontinued	<code>lcl.datetime.date.digitFormatEnable</code>	User preference parameter
Discontinued	<code>log.level.change.lp</code>	Log parameter
Discontinued	<code>log.level.change.nwmgr</code>	Log parameter
Discontinued	<code>log.level.change.sync</code>	Log parameter
Discontinued	<code>reg.x.filterReflectedBlaDialogs</code>	Call feature parameter

<i>Modification</i>	<i>Configuration Parameter</i>	<i>Description</i>
Discontinued	<code>reg.x.server.H323.y.register</code>	Call feature parameter
Discontinued	<code>sec.TLS.customDeviceCert.enable</code>	Security parameter
Discontinued	<code>sec.dot1x.eapollogoff.pcforselanlinkreset</code>	Security parameter
Discontinued	<code>voIpProt.SDP.useLegacyPayloadType Negotiation</code>	Call feature parameter
Discontinued	<code>voIpProt.server.H323.x.register</code>	Call feature parameter
Discontinued	<code>voice.aec.hd.*</code> The parameter <code>voice.aec.hd.enable</code> has not been removed	Audio parameters
Discontinued	<code>voice.aec.hf.*</code> The parameter <code>voice.aec.hf.enable</code> has not been removed	Audio parameters
Discontinued	<code>voice.aec.hs.*</code> The parameter <code>voice.aec.hs.enable</code> has not been removed	Audio parameters
Discontinued	<code>voice.aes.hd.duplexBalance</code>	Audio parameter
Discontinued	<code>voice.aes.hf.*</code> The parameter <code>voice.aes.hf.enable</code> has not been removed	Audio parameters
Discontinued	<code>voice.aes.hs.duplexBalance</code>	Audio parameter
Discontinued	<code>voice.gain.rx.digital.ringer.*</code> The parameter <code>voice.gain.rx.digital.ringer</code> has also been removed	Audio parameters
Discontinued	<code>voice.handset.wideband</code>	Audio parameter
Discontinued	<code>voice.rxEq.hf.postFilter.*</code> The parameter <code>voice.rxEq.hf.postFilter.enable</code> has not been removed	Audio parameters
Discontinued	<code>voice.rxEq.hf.preFilter.*</code> The parameter <code>voice.rxEq.hf.preFilter.enable</code> has not been removed	Audio parameters
Discontinued	<code>voice.rxEq.hs.postFilter.*</code> The parameter <code>voice.rxEq.hs.postFilter.enable</code> has not been removed	Audio parameters
Discontinued	<code>voice.rxEq.hs.preFilter.*</code> The parameter <code>voice.rxEq.hs.preFilter.enable</code> has not been removed	Audio parameters

<i>Modification</i>	<i>Configuration Parameter</i>	<i>Description</i>
Added	apps.x.label	Productivity Applications parameter
Added	apps.x.url	Productivity Applications parameter
Added	apps.push.secureTunnelEnabled	Productivity Applications parameter
Added	apps.push.secureTunnelPort	Productivity Applications parameter
Added	apps.push.secureTunnelRequired	Productivity Applications parameter
Added	apps.statePolling.responseMode	Productivity Applications parameter
Added	apps.telNotification.callStateChangeEvent	Productivity Applications parameter
Added	apps.telNotification.userLogInOutEvent	Productivity Applications parameter
Added	apps.ucdesktop.* The parameter apps.ucdesktop.enabled has existed in previous versions	Productivity Applications parameters
Added	bg.color.*	Background parameter
Added	bluetooth.radioOn	Call feature parameter
Added	call.advancedMissedCalls.*	Call feature parameters
Added	call.callWaiting.enable	Call feature parameter
Added	call.localConferenceEnabled	Call feature parameter
Added	callLists.*	Call feature parameters
Added	device.hostname	Provisioning parameter
Added	device.net.dhcpBootServer	Provisioning parameter
Added	device.net.dot1x.*	Provisioning parameters
Added	device.pacfile.*	Provisioning parameters
Added	device.prov.upgradeServer	Provisioning parameter
Added	device.sec.TLS.*	Provisioning parameters
Added	device.usbnet.*	Provisioning parameters
Added	device.wifi.*	Provisioning parameter
Added	dialplan.applyToPstnDialing	Call feature parameter
Added	dialplan.routing.emergency.outboundIdentity	Call feature parameter
Added	dialplan.routing.emergency.outboundIdentity.lldp	Call feature parameter

<i>Modification</i>	<i>Configuration Parameter</i>	<i>Description</i>
Added	dialplan.routing.ermgency.preferredSource	Call feature parameter
Added	exchange.meeting.*	Productivity Applications parameter
Added	exchange.server.url	Productivity Applications parameter
Added	feature.audioVideoToggle.enabled	Call feature parameter
Added	feature.bluetooth.enabled	Call feature parameter
Added	feature.enhancedCallDisplay.enabled	Call feature parameter
Added	feature.exchangeCalendar.enabled	Call feature parameter
Added	feature.nonVolatileRingerVolume.enabled	Call feature parameter
Added	httpd.cfg.secureTunnelEnabled	Web Configuration Utility parameter
Added	httpd.cfg.secureTunnelPort	Web Configuration Utility parameter
Added	httpd.cfg.secureTunnelRequired	Web Configuration Utility parameter
Added	httpd.ta.secureTunnelEnabled	Web Configuration Utility parameter
Added	httpd.ta.secureTunnelPort	Web Configuration Utility parameter
Added	httpd.ta.secureTunnelRequired	Web Configuration Utility parameter
Added	ind.pattern.blink.*	LED indicator parameter
Added	ind.pattern.flashSlow2.*	LED indicator parameter
Added	lcl.x.pstnCountry	Multilingual parameter
Added	lcl.aidt	Multilingual parameter
Added	lcl.callerId	Multilingual parameter
Added	lcl.callerIdType	Multilingual parameter
Added	lcl.country.* The parameter lcl.country has also been added	Multilingual parameters
Added	lcl.dtmfLevel	Multilingual parameter
Added	lcl.dtmfTwist	Multilingual parameter
Added	lcl.flashTiming	Multilingual parameter
Added	lcl.pstnCountryIndex	Multilingual parameter
Added	lineKey.*	Flexible line key assignment parameters
Added	log.level.change.barcd	Log parameter

<i>Modification</i>	<i>Configuration Parameter</i>	<i>Description</i>
Added	log.level.change.bluet	Log parameter
Added	log.level.change.clist	Log parameter
Added	log.level.change.daa	Log parameter
Added	log.level.change.dock	Log parameter
Added	log.level.change.drvtb	Log parameter
Added	log.level.change.oaip	Log parameter
Added	log.level.change.ocsp	Log parameter
Added	log.level.change.pdc	Log parameter
Added	log.level.change.pres	Log parameter
Added	log.level.change.pstn	Log parameter
Added	log.level.change.ptt	Log parameter
Added	log.level.change.rtls	Log parameter
Added	log.level.change.tls	Log parameter
Added	log.level.change.wifi	Log parameter
Added	log.render.stdout.* The parameter log.render.stdout has existed in previous versions	Log parameter
Added	mb.main.toolbar.autoHide.* The parameter mb.main.toolbar.autoHide.enabled has existed in previous versions	Microbrowser parameters
Added	messaging.*	SpectraLink instant messaging parameters
Added	np.*	SpectraLink notification profiles parameters
Added	oai.*	SpectraLink Open Application Interface parameters
Added	prov.login.*	Distributed polling parameters
Added	prov.loginCredPwdFlushed.enabled	Distributed polling parameter
Added	prov.polling.timeRandomEnd	Distributed polling parameter
Added	prov.startupCheck.enabled	Provisioning parameter
Added	pstn.*	

<i>Modification</i>	<i>Configuration Parameter</i>	<i>Description</i>
Added	ptt.*	Paging and push-to-talk parameters
Added	qbc.*	SpectralLink quick barcode connector parameters
Added	qos.ethernet.* The parameters qos.ethernet.callControl.user_priority, qos.ethernet.other.user_priority, qos.ethernet.rtp.user_priority, and qos.ethernet.rtp.video.user_priority exist in previous versions	SpectralLink QoS parameters
Added	reg.x.auth.domain	Call feature parameter
Added	reg.x.auth.useLoginCredentials	Call feature parameter
Added	reg.x.gruu	Call feature parameter
Added	reg.x.server.y.specialInterop	Call feature parameter
Added	reg.x.server.y.useOutboundProxy	Call feature parameter
Added	reg.x.srtp.enable	Call feature parameter
Added	reg.x.srtp.offer	Call feature parameter
Added	reg.x.srtp.require	Call feature parameter
Added	reg.x.telephony	Call feature parameter
Added	se.pat.misc.customX.*	Sound effects parameters
Added	se.pat.misc.miscX.*	Sound effects parameters
Added	se.rt.answerMute.*	Sound effects parameters
Added	se.rt.autoAnswer.micMute	Sound effects parameter
Added	se.rt.autoAnswer.videoMute	Sound effects parameter
Added	se.rt.customX.micMute	Sound effects parameter
Added	se.rt.customX.videoMute	Sound effects parameter
Added	se.rt.default.micMute	Sound effects parameter
Added	se.rt.default.videoMute	Sound effects parameter
Added	se.rt.emergency.micMute	Sound effects parameter
Added	se.rt.emergency.videoMute	Sound effects parameter
Added	se.rt.external.micMute	Sound effects parameter

<i>Modification</i>	<i>Configuration Parameter</i>	<i>Description</i>
Added	se.rt.external.videoMute	Sound effects parameter
Added	se.rt.internal.micMute	Sound effects parameter
Added	se.rt.internal.videoMute	Sound effects parameter
Added	se.rt.precedence.micMute	Sound effects parameter
Added	se.rt.precedence.videoMute	Sound effects parameter
Added	se.rt.ringAnswerMute.*	Sound effects parameters
Added	se.rt.splash.micMute	Sound effects parameter
Added	se.rt.splash.videoMute	Sound effects parameter
Added	se.rt.visual.micMute	Sound effects parameter
Added	se.rt.visual.videoMute	Sound effects parameter
Added	sec.TLS.SIP.strictCertCommonNameValidation	Security parameter
Added	sec.TLS.customCaCert.*	Security parameters
Added	sec.TLS.customDeviceCert.*	Security parameters
Added	sec.TLS.customDeviceKey.*	Security parameters
Added	sec.TLS.profile.*	Security parameters
Added	sec.TLS.profileSelection.*	Security parameters
Added	sec.hostMoveDetect.*	Security parameters
Added	sec.srtp.holdWithNewKey	Security parameter
Added	sec.srtp.mki.length	Security parameter
Added	sec.srtp.resumeWithNewKey	Security parameter
Added	softkey.x.insert	Security parameter
Added	tcpIpApp.fileTransfer.waitForLinkIfDown	IP parameter
Added	tone.chord.misc.A3Major.*	Tone parameters
Added	tone.chord.misc.C3Major.*	Tone parameters
Added	tone.chord.misc.Db3Major.*	Tone parameters
Added	tone.chord.misc.E3Major.*	Tone parameters
Added	tone.chord.misc.cs12.*	Tone parameters
Added	up.25mmRealTime	User preferences parameter

<i>Modification</i>	<i>Configuration Parameter</i>	<i>Description</i>
Added	up.backlight.timeout.* The parameter up.backlight.timeout has also been added	User preferences parameters
Added	up.cfgWarningsEnabled	User preferences parameter
Added	up.displayOperMode	User preferences parameter
Added	up.headsetOnlyAlerting	User preferences parameter
Added	up.hearingAidCompatibility.enabled	User preferences parameter
Added	up.hideDateTimeWhenNotSet	User preferences parameter
Added	up.multiKeyAnswerEnabled	User preferences parameter
Added	up.operMode	User preferences parameter
Added	up.pstnSetup	User preferences parameter
Added	up.warningLevel	User preferences parameter
Added	upgrade.*	Provisioning parameter
Added	video.callMode.default	Video parameter
Added	video.debug	Video parameter
Added	voIpProt.SIP.dialog.strictXLineId	Call feature parameter
Added	voIpProt.SIP.IM.autoAnswerDelay	Call feature parameter
Added	voIpProt.SIP.mtls.enable	Call feature parameter
Added	voIpProt.SIP.pingMethod	Call feature parameter
Added	voIpProt.server.x.specialInterop	Call feature parameter
Added	voIpProt.server.x.useOutboundProxy	Call feature parameter
Added	voice.aec.bt.hd.enable	Audio parameter
Added	voice.aec.usb.hf.enable	Audio parameter
Added	voice.aes.bt.hd.enable	Audio parameter
Added	voice.aes.usb.hf.enable	Audio parameter
Added	voice.agc.bt.hd.enable	Audio parameter
Added	voice.agc.usb.hf.enable	Audio parameter
Added	voice.bt.*	Audio parameters
Added	voice.handset.rxag	Audio parameter

<i>Modification</i>	<i>Configuration Parameter</i>	<i>Description</i>
Added	voice.handset.rxdg	Audio parameter
Added	voice.handset.st.	Audio parameter
Added	voice.handset.txag	Audio parameter
Added	voice.handset.txdg	Audio parameter
Added	voice.handsfree.*	Audio parameters
Added	voice.headset.rxag	Audio parameter
Added	voice.headset.rxdg	Audio parameter
Added	voice.headset.st	Audio parameter
Added	voice.headset.txag	Audio parameter
Added	voice.headset.txdg	Audio parameter
Added	voice.ns.bt.*	Audio parameters
Added	voice.ns.usb.*	Audio parameters
Added	voice.ringer.rxag	Audio parameter
Added	voice.rxEq.usb.*	Audio parameters
Added	voice.rxQos.ptt.*	Audio parameters
Added	voice.rxQos.wireless.*	Audio parameters
Added	voice.txEq.usb.*	Audio parameters
Added	voice.usb.*	Audio parameters
Added	voice.volume.persist.bluetooth.headset	Audio parameter
Added	voice.volume.persist.usb.handsfree	Audio parameter
Added	wifi.*	Wifi parameters
Changed	apps.push.messageType	Productivity Applications parameter
Changed	apps.uc.desktop.enabled	Productivity Applications parameter
Changed	call.autoRouting.preferredProtocol	Call feature parameter
Changed	device.auth.*	Provisioning parameters
Changed	device.cma.mode	Provisioning parameter
Changed	device.dhcp.*	Provisioning parameters
	The parameter device.dhcp.bootSrvOpt has not been changed	

<i>Modification</i>	<i>Configuration Parameter</i>	<i>Description</i>
Changed	device.dns.*	Provisioning parameters
Changed	device.em.power	Provisioning parameter
Changed	device.line.*	Provisioning parameters
Changed	device.net.* The parameters device.net.dhcpBootServer, device.net.IPgateway, device.net.subnetMask, and device.net.vlanId have not been changed	Provisioning parameters
Changed	device.ntlm.versionMode	Provisioning parameter
Changed	device.prov.* The parameters device.prov.password, device.prov.serverName, device.prov.upgradeServer, and device.prov.user have not been changed	Provisioning parameters
Changed	device.serial.enable	Provisioning parameter
Changed	device.sntp.gmtOffset	Provisioning parameter
Changed	device.syslog.* The parameter device.syslog.serverName has not been changed	Provisioning parameters
Changed	dialplan.x.digitmap	Call feature parameter
Changed	dialplan.x.routing.server.y.transport	Call feature parameter
Changed	dialplan.digitmap	Call feature parameter
Changed	dialplan.digitmap.timeOut	Call feature parameter
Changed	dialplan.impossibleMatchHandling	Call feature parameter
Changed	dialplan.removeEndOfDial	Call feature parameter
Changed	dialplan.routing.server.x.transport	Call feature parameter
Changed	dir.H350.dev.attribute.x.type	Directory parameter
Changed	dir.H350.dev.transport	Directory parameter
Changed	dir.H350.group.attribute.x.type	Directory parameter
Changed	dir.H350.group.transport	Directory parameter
Changed	dir.H350.person.attribute.x.type	Directory parameter
Changed	dir.H350.person.transport	Directory parameter
Changed	dir.corp.attribute.x.type	Directory parameter

<i>Modification</i>	<i>Configuration Parameter</i>	<i>Description</i>
Changed	<code>dir.corp.transport</code>	Directory parameter
Changed	<code>dir.local.nonVolatile.maxSize</code>	Directory parameter
Changed	<code>dir.local.volatile.maxSize</code>	Directory parameter
Changed	<code>divert.x.autoOnSpecificCaller</code>	Call feature parameter
Changed	<code>divert.x.contact</code>	Call feature parameter
Changed	<code>divert.x.sharedDisabled</code>	Call feature parameter
Changed	<code>divert.busy.*</code>	Call feature parameters
Changed	<code>divert.dnd.*</code>	Call feature parameters
Changed	<code>divert.fwd.x.enabled</code>	Call feature parameter
Changed	<code>divert.noanswer.*</code>	Call feature parameters
Changed	<code>ind.led.x.index</code>	LED indicator parameter
Changed	<code>keypadLock.*</code>	Phone lock parameters
Changed	<code>lcl.ml.lang.clock.x.format</code>	Multilingual parameter
Changed	<code>lcl.ml.lang.list</code>	Multilingual parameter
Changed	<code>lcl.ml.lang.menu.*</code>	Multilingual parameters
Changed	<code>lcl.ml.lang.tags.*</code>	Multilingual parameters
Changed	<code>log.level.change.slog</code>	Log parameter
Changed	<code>log.render.file.size</code>	Log parameter
Changed	<code>log.render.file.upload.append.limitMode</code>	Log parameter
Changed	<code>log.render.file.upload.period</code>	Log parameter
Changed	<code>log.render.stdout</code>	Log parameter
Changed	<code>log.sched.*</code>	Log parameters
Changed	<code>msg.bypassInstantMessage</code>	Voicemail parameter
Changed	<code>nat.*</code> The parameter <code>nat.keepalive.interval</code> has not been changed	IP parameters
Changed	<code>phoneLock.enabled</code>	Phone lock parameter
Changed	<code>pnet.remoteCall.dtmfDuration</code>	Peer networking parameter
Changed	<code>powerSaving.enable</code>	Power saving parameter

<i>Modification</i>	<i>Configuration Parameter</i>	<i>Description</i>
Changed	prov.fileSystem.ffs0.minFreeSpace	Provisioning parameter
Changed	prov.polling.mode	Distributed polling parameter
Changed	prov.polling.time	Distributed polling parameter
Changed	qos.ethernet.*	Quality of Service parameters
Changed	qos.ip.callControl.* The parameters qos.ip.callControl.dscp.* have not been changed	Quality of Service parameters
Changed	qos.ip.rtp.* The parameters qos.ip.rtp.dscp.* and qos.ip.rtp.video.dscp.* have not been changed	Quality of Service parameters
Changed	reg.x.callsPerLineKey	Line registration parameter
Changed	reg.x.outboundProxy.transport	Line registration parameter
Changed	reg.x.ringType	Line registration parameter
Changed	reg.x.server.y.transport	Line registration parameter
Changed	res.finder.minFree	Phone memory parameter
Changed	res.finder.sizeLimit	Phone memory parameter
Changed	res.quotas.background	Phone memory parameter
Changed	res.quotas.bitmap	Phone memory parameter
Changed	res.quotas.cache	Phone memory parameter
Changed	res.quotas.font	Phone memory parameter
Changed	res.quotas.tone	Phone memory parameter
Changed	res.quotas.xmlui	Phone memory parameter
Changed	roaming_buddies.reg	SpectralLink call feature parameter
Changed	roaming_privacy.reg	SpectralLink call feature parameter
Changed	se.destination	Sound effects parameter
Changed	se.pat.callProg.msgWaiting.name	Sound effects parameter
Changed	se.pat.misc.instantMessage.name	Sound effects parameter
Changed	se.pat.misc.localHoldNotification.name	Sound effects parameter
Changed	se.pat.misc.messageWaiting.name	Sound effects parameter
Changed	se.pat.misc.negativeConfirm.name	Sound effects parameter

<i>Modification</i>	<i>Configuration Parameter</i>	<i>Description</i>
Changed	se.pat.misc.positiveConfirm.name	Sound effects parameter
Changed	se.pat.misc.remoteHoldNotification.name	Sound effects parameter
Changed	se.pat.misc.welcome.name	Sound effects parameter
Changed	sec.H235.*	Sound effects parameters
Changed	tcpIpApp.keepalive.*	IP parameters
Changed	tcpIpApp.port.*	IP parameters
Changed	up.25mm	User preferences parameter
Changed	up.analogHeadsetOption	User preferences parameter
Changed	up.backlight.idleIntensity	User preferences parameter
Changed	up.oneTouchVoiceMail	User preferences parameter
Changed	up.useDirectoryNames	User preferences parameter
Changed	up.welcomeSoundEnabled	User preferences parameter
Changed	up.welcomeSoundOnWarmBootEnabled	User preferences parameter
Changed	video.camera.frameRate	Video parameter
Changed	video.localCameraView.fullScreen.mode	Video parameter
Changed	video.maxCallRate	Video parameter
Changed	video.screenMode	Video parameter
Changed	video.screenModeFS	Video parameter
Changed	voIpProt.H323.dtmfViaSignaling	Call feature parameter
Changed	voIpProt.H323.enable	Call feature parameter
Changed	voIpProt.H323.local.port	Call feature parameter
Changed	voIpProt.SIP.local.port	Call feature parameter
Changed	voIpProt.SIP.outboundProxy.transport	Call feature parameter
Changed	voIpProt.SIP.specialEvent.lineSeize.nonStandard	Call feature parameter
Changed	voIpProt.server.x.transport	Call feature parameter
Changed	voIpProt.server.dhcp.option	Call feature parameter
Changed	voice.audioProfile.Lin16.16ksps.payloadSize	Audio parameter

<i>Modification</i>	<i>Configuration Parameter</i>	<i>Description</i>
Changed	voice.audioProfile.Lin16.32ksps.payloadSize	Audio parameter
Changed	voice.audioProfile.Lin16.44_1ksps.payloadSize	Audio parameter
Changed	voice.audioProfile.Lin16.48ksps.payloadSize	Audio parameter
Changed	voice.audioProfile.Lin16.8ksps.payloadSize	Audio parameter
Changed	voice.codecPref.iLBC.*	Audio parameters
Changed	voice.gain.rx.analog.*	Audio parameters
Changed	voice.gain.rx.digital.*	Audio parameters
Changed	voice.gain.tx.analog.*	Audio parameters
Changed	voice.gain.tx.digital.*	Audio parameters
Changed	voice.handset.rxag.adjust.*	Audio parameters
Changed	voice.handset.sidetone.adjust.*	Audio parameters
Changed	voice.handset.txag.adjust.*	Audio parameters
Changed	voice.headset.sidetone.adjust.*	Audio parameters
Changed	voice.headset.txag.adjust*	Audio parameters
Changed	voice.ns.hd.* The parameter voice.ns.hd.enable has not been changed	Audio parameters
Changed	voice.ns.hf.signalAttn	Audio parameter
Changed	voice.ns.hf.silenceAttn	Audio parameter
Changed	voice.ns.hs.signalAttn	Audio parameter
Changed	voice.ns.hs.silenceAttn	Audio parameter

Reference Documents

This section lists all documents referenced in these release notes and other relevant documents.

For information and support for all Polycom voice products and software, and for access to supporting documentation, see [Polycom UC Software Support Center](#).

For additional information about the Polycom SoundPoint IP, SoundStationIP, SoundStation Duo, VVX Business Media Phones, the VVX Camera, and the VVX Expansion Modules, view the following support pages:

- [Polycom SoundPoint IP](#)
- [Polycom SoundStation IP](#)
- [Polycom SoundStation Duo](#)
- [Polycom VVX 300 and 310](#)
- [Polycom VVX 400 and 410](#)
- [Polycom VVX 500](#)
- [Polycom VVX 600](#)
- [Polycom VVX 1500](#)
- [Polycom VVX Camera](#)
- [Polycom VVX Expansion Modules](#)

You can view the following types of documents on each product page:

User Documents:

➤ *Quick User Guide* A quick reference on how to use the phone's most basic features.

• Setup and Maintenance Documents:

➤ *Quick Start Guide* This guide describes the contents of your package, how to assemble the phone or accessory, and how to connect the phone to the network. The quick start guide is included in your phone package.

➤ *Wallmount Instructions* This document provides detailed instructions for mounting your phone on the wall. To install your phone on the wall, you need the optional wallmount package, which includes the wallmount instructions.

➤ *Administrator's Guide* This guide provides detailed information about setting up your network and configuring phone features.

Feature Descriptions and Technical Notifications These documents describe workarounds to existing issues and provide expanded descriptions and examples for phone settings and features. You can find these documents on the [Polycom Profiled UC Software Features](#) and [Polycom Engineering Advisories and Technical Notifications](#) support pages.

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