



#CiscoLive

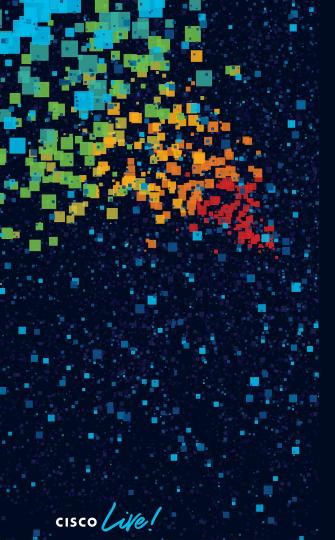
## Cisco IP Phone Update

Best Practices and Collaboration Enablement

David Scott CCIE, Technical Marketing Engineer DGTL-BRKUCC-2050

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### Agenda

Part 1 - IP Phone Update

- Remote Working Update
- Simple Phone Refresh
- 7800/8800 Update
- Multiplatform Phones

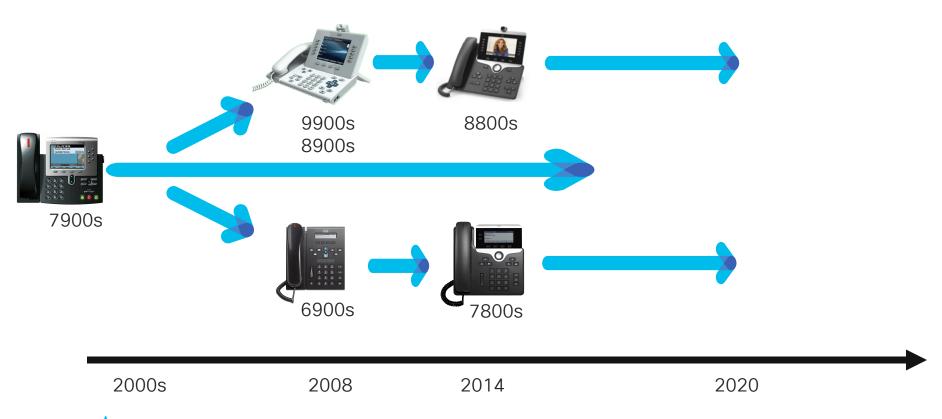
Part 2 - Headset Update

Part 3 - Conference Phone Update

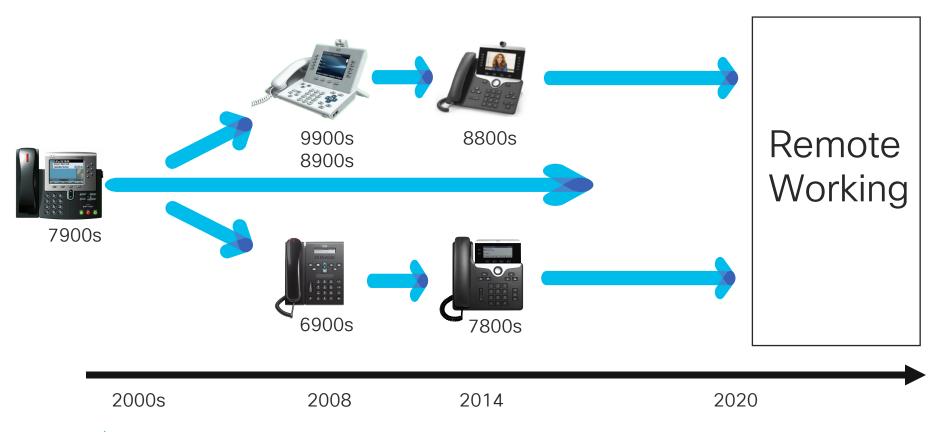
Webex Room Phone



## A Brief History of Cisco IP Phones



## A Brief History of Cisco IP Phones



#### Remote Work Experiences

Best



Better



Good



Components:

- Webex Desk Pro or DX80
- Cisco Headset 500 or 700
- Laptop: Jabber, Webex

Components:

- Cisco IP Phone (Ent or MPP) •
- Cisco Headset 500 or 700
- Laptop: Jabber, Webex

Components:

- Cisco Headset 500 or 700
- Laptop: Jabber, Webex



#### Cisco IP Phone 7800 Series

**Key Features** 



CP-7821

CP-7841

CP-7861









Replaceable Bezel				
Screen	384 x 106 Mono (3.28")	396 x 162 Mono (3.5")	396 x 162 Mono (3.5")	396 x 162 Mono (3.5")
Ethernet Switch	10/100	10/100	10/100/1000	10/100
Programmable Line Keys	N/A	2	4	16
Programmable Soft Keys	4	4	4	4
Headset Port	N/A			
EHH Support (AUX)	N/A			
Full Duplex Speakerphone	(Narrowband)			
Wideband Audio	Available with additional wideband handset			
POE Class	1	1	1	1
Protocols	SIP	SIP	SIP	SIP



### Cisco IP Phone 8800 Series

Key Features











Display	800 x 480 Greyscale(5")	800 x 480 WVGA (5")				
Ethernet Switch	10 / 100 / 1000	10 / 100 / 1000	10 / 100 / 1000	10 / 100 / 1000	10 / 100 / 1000	10 / 100 / 1000
Programmable Line Keys	5 x 5	5 x 5	5 x 5	5 x 5	5 x 5	5 x 5
Programmable Soft Keys	4	4	4	4	4	4
Headset Port (RJ9)						
EHS Support						
Full Duplex Speakerphone						
Wideband Audio						
Integrated Bluetooth						
USB						
KEM Support				(2)	(3)	(3)
Wifi						
External Audio Port						
HD 720p Video						
POE Class	2	2	2	4	4	4



## Cisco IP Phone: Power Supply and Injectors

Power Options	6821/51/71	79xx	78xx	88xx (excludes 8821/8832)
		(802.3af)*	(excludes 7832)	
CP-6800-PWR	Yes	No	No	No
CP-PWR-CUBE-3*	No	Yes	Yes	No
CP-PWR-ADPT-3	No	No	Yes (Recommended)	No
CP-PWR-CUBE-4	No	No	No	Yes (KEMs supported)
AIR-PWRINJ5 (Class 3 power injector)	Yes	Yes	Yes	8811/41/45/51/61/65 (No KEM support)**
AIR-PWRINJ6 (Class 4 power injector)	Yes	Yes	Yes	8811/41/45/51/61/65 (No KEM support)**
Meraki MA-INJ-4-US	Yes	Yes	Yes	8811/41/45/51/61/65 (No KEM support)**

<sup>\*</sup> Some older 7900 phones do not support CUBE-3 or may only support pre-standard POE. http://cs.co/pre-standard \*\*AIR-PWRINJ6 and MA-INJ-4-US support 802.3at, and the 8851(v08), 8861 and 8865 are IEEE Class 4 devices, yet the phones require CDP or LLDP protocol negotiation in order to power up a KEM. Injectors do not provide CDP or LLDP.

## Power Adaptor 3 for 78xx series



Lower Input Power Consumption

**Smaller Size** 

Verified for Cisco Collaboration Endpoint Fully compatible with 78x1 & 7832 (both OnPrem & MPP) Worldwide voltage/frequency support

Continued till End of CY2020



## Power Adaptor 3 for 78xx series (continued)

PID	Regions
CP-PWR-ADPT-3-NA=	North America/Japan
CP-PWR-ADPT-3-EU=	EU/Switzerland
CP-PWR-ADPT-3-AU=	Australia/New Zealand
CP-PWR-ADPT-3-UK=	United Kingdom/Asia Pacific
CP-PWR-ADPT-3-CN=	China
CP-PWR-ADPT-3-IN=	India
CP-PWR-ADPT-3-KR=	Korea
CP-PWR-ADPT-3-BZ=	Brazil
CP-PWR-ADPT-3-AR=	Argentina

(Clip already included in PID)





Webex Room 55/70 MRA Client & Endpoint Support Webex Room Kit Webex Room Kit Plus Available Today Webex Room Kit Pro Webex Room Kit Mini External Network Internal Network DMZ DX650, DX70, DX80 Internet MX, SX, EX, C Series Expressway-C Expressway-E TelePresence Endpoints 8811, 8841, 8845, 8851, 8861, 8865 7832 & 8832

7811, 7821, 7841, 7861

## Activation Code Onboarding for CUCM

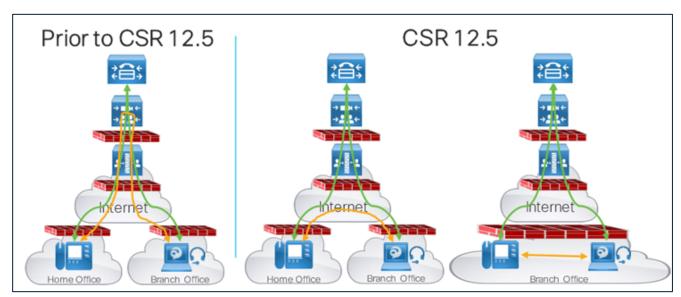
- This new functionality provides assurance that a new or untrusted phone seen on the network cannot register with Cisco Unified Call Manager without first having an activation code and having its internal Manufacturing Installed Certificate (MIC) verified
  - Allows for phones to be onboarded in a bulk fashion without requiring the use of auto-registration
  - Supported phone models in CUCM 12.5 SU1 7811, 7821, 7832, 7841, 7861, 8811, 8832, 8841, 8845, 8851, 8851NR, 8861, 8865, and 8865NR
  - MRA Onboarding Planned Dependencies: Expressway X12.5.3 + CUCM 12.5 SU1 + 78xx/88xx 12.5(1)SR3
  - See BRKUCC-2801 Enabling External Collaboration with Expressway





# Media Path Optimization for MRA Calls

- Requires 12.5 or higher phone firmware on 7800/8800
- Requires CUCM 12.5 + Expressway X12.5







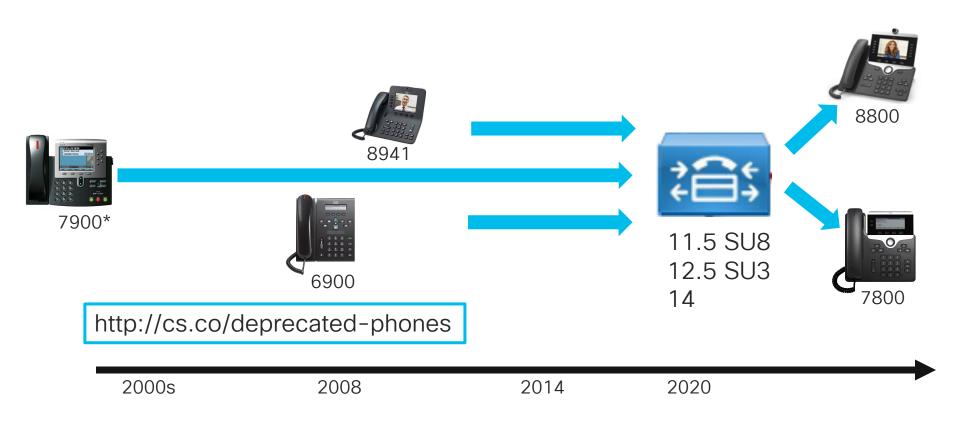
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These statements are based on estimates and information. available to us at the time of this presentation and are not quarantees of future performance. Actual results could differ materially from our current expectations as a results of many factors, including but not limited to: the unpredictable nature of our rapidly evolving market and quarterly fluctuations in our business; the effects of competition; and any adverse changes in our indirect channel relationships. These and other risks and uncertainties associated with our business are described in the company's annual report on Form 10-K. The forward-looking statements in this presentation are made as of the date of the initial publication of this presentation. and we disclaim any obligation to update these statements at any time in the future.



## A Short History of Cisco IP Phones



<sup>\*</sup> https://www.cisco.com/c/en/us/td/docs/voice\_ip\_comm/cucm/trouble/14\_0\_1/fieldNotices/cucm\_b\_deprecated-phones-14.html

## Phone refresh simplified http://cs.co/simple-phone-refresh

An easy, intuitive Cisco IP phone migration that minimizes the complexity, cost, and time for professional services and is a native solution in CUCM.



- Unplug a deprecated/faulty phone from the network
- 2 Plug-in a replacement phone
- 3 Enter Primary DN, (optionally) PIN or authorization code on the new phone
- 4 New phone with all the settings from the old phone is provisioned







# Simple Phone Refresh CUCM 11.5 SU8, CUCM 12.5 SU3, CUCM 14

	Phone migration service (non-IVR)	Self-provisioning IVR	CM administration interface
End user or administrator driven phone migration	End user driven (Self-service)	End user driven (Self-service)	Administrator driven
Auto-registration required	No	Yes	No
Migration steps	<ol> <li>Plug-in new phone to the network</li> <li>Key in primary extension and PIN</li> </ol>	<ol> <li>Auto register a new phone</li> <li>Dial self-provisioning IVR number and follow voice prompts</li> </ol>	<ol> <li>Sign in to CM         Administration interface</li> <li>Choose "Migrate         Phone" option in Phone         Configuration page of         old phone</li> <li>Enter phone type         (model &amp; protocol) and         MAC address of the         new phone</li> </ol>
Administrator involvement	Low	Medium	High



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Administrator involvement	Low	Medium	High



# Phone Migration Service (non-IVR) End User Experience

1 Unplug a deprecated or faulty phone from the network



2 Unbox and plug in new phone





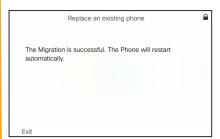
Phone downloads a new firmware that supports simple phone refresh

Enter Primary DN, (Optionally) PIN





4 Migration Successful





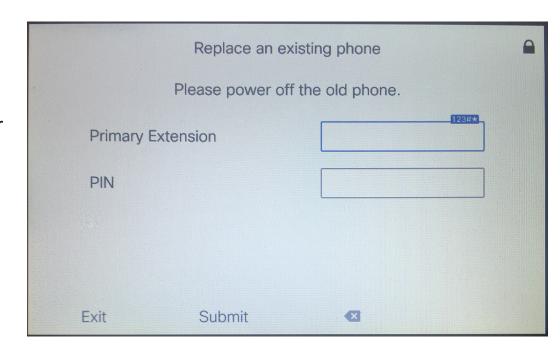
New phone with all the settings from the old phone is provisioned



### Phone Migration Service (non-IVR) CUCM 11.5 SU8, CUCM 12.5 SU3, CUCM 14

#### How it works:

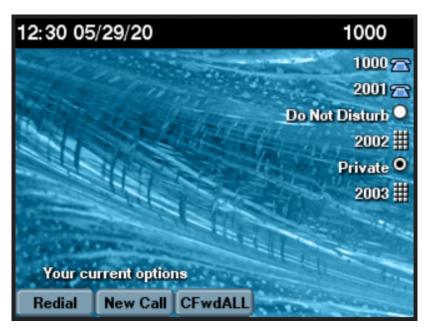
- Phone Migration Service is always available and cannot be started/stopped
- 7800 and 8800 need to run 12.8 or higher
- Auto-registration needs to be disabled
- Self Provisioning IVR service is NOT used, but its settings can be used
  - End user can replace their own phone
  - Admin can replace on behalf of end user

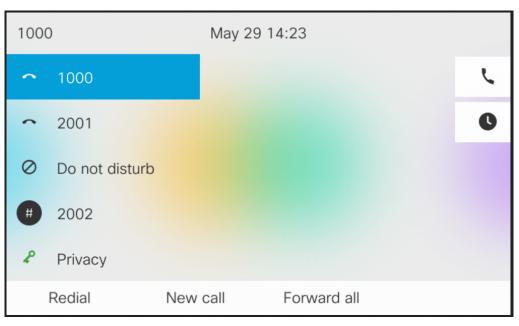




# Phone Migration Example: 7900 to 8800 CUCM 11.5 SU8, CUCM 12.5 SU3, CUCM 14

7975 8861







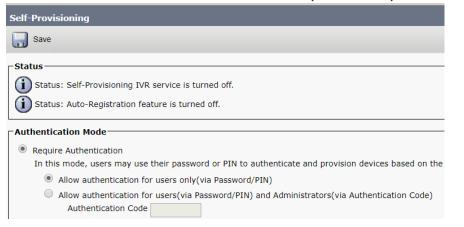
<sup>\*</sup>Phone will move to Session Line Mode by default

#### Phone Migration Service (non-IVR) CUCM 11.5 SU8, CUCM 12.5 SU3, CUCM 14

- Self Provisioning IVR service is NOT used, but its settings page is used for non-IVR authentication
  - End user can replace their own phone with PIN
  - Admin can replace for end user with Admin PIN
  - Admin can select no PIN required (no auth)



#### Allow End User PIN (default)



#### Allow Fnd User PIN and Admin PIN

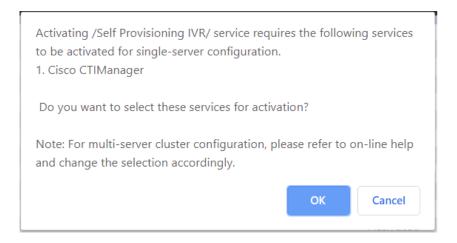
Self-Provisioning
Save
_Status—
Status: Self-Provisioning IVR service is turned off.
Status: Auto-Registration feature is turned off.
Authentication Mode
Require Authentication
In this mode, users may use their password or PIN to authenticate and provision devices based on the
Allow authentication for users only(via Password/PIN)
<ul> <li>Allow authentication for users(via Password/PIN) and Administrators(via Authentication Code)</li> <li>Authentication Code</li> </ul>

# Simple Phone Refresh CUCM 11.5 SU8, CUCM 12.5 SU3, CUCM 14

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Administrator involvement	Low	Medium	High



## Phone Migration with Self-provisioning IVR CUCM 11.5 SU8, CUCM 12.5 SU3, CUCM 14



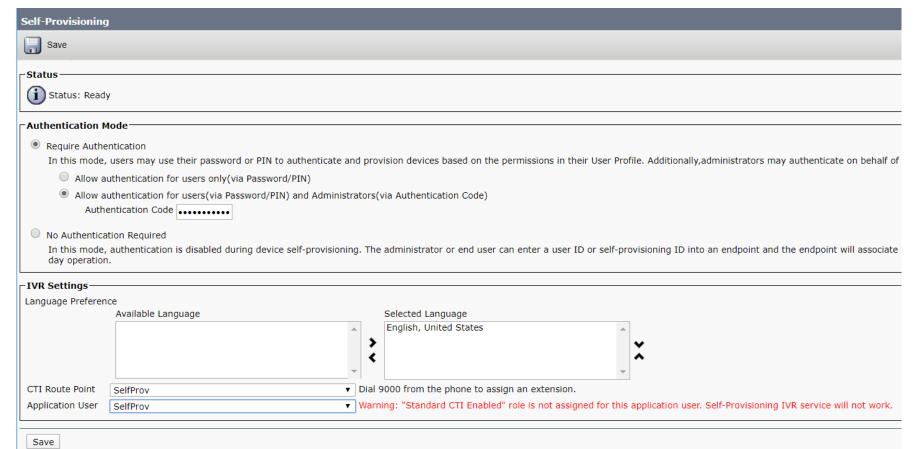
CTI Services			
	Service Name	Activation Status	
	Cisco IP Manager Assistant	Deactivated	
	Cisco WebDialer Web Service	Deactivated	
	Self Provisioning IVR	Deactivated	

## Self-Provisioning IVR – Configuration Steps CUCM 11.5 SU8, CUCM 12.5 SU3, CUCM 14

- Enable Auto-registration.
- End users must have a primary extension.
- End users must be associated to a user profile or feature group template that includes a universal line template, universal device template and which has Self-Provisioning enabled.
- Ensure that the right "CTI Route Point" and "Application User" configurations are selected.
- Enable Self-Provisioning IVR service.



## Self-Provisioning CUCM page



## Self-Provisioning Process CUCM 11.5 SU8, CUCM 12.5 SU3, CUCM 14

#### Before you begin

Make sure that the old phone is in the "Unregistered" state before you proceed with migration. You can plug the new phone into the network, wait until the phone registers and then perform the migration tasks. Once the migration is successful, the device will re-register with the users phone configuration data.

#### **Procedure**

- Step 1 Dial the extension that is assigned to the Self-Provisioning IVR from a new Cisco IP Phone or device.
- Step 2 Press 2 for phone replacement option.
- Step 3 Enter the primary extension number of your phone followed by the pound key (#).
- Step 4 Enter the end user PIN followed by the pound key (#).

Migration starts after a successful authentication. Following the migration, the phone restarts with the same configuration settings.

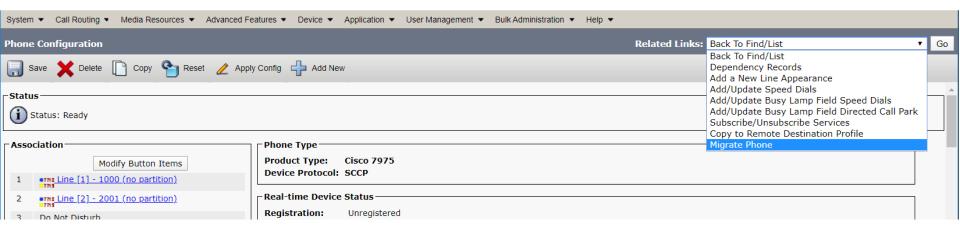


# Simple Phone Refresh CUCM 12.5 SU3 & CUCM 11.5 SU8

	Phone migration service (non-IVR)	Self-provisioning IVR	CM administration interface
End user or administrator driven phone migration	End user driven (Self-service)	End user driven (Self-service)	Administrator driven
Auto-registration required	No	Yes	No
Migration steps	<ol> <li>Plug-in new phone to the network</li> <li>Key in primary extension and PIN</li> </ol>	<ol> <li>Auto register a new phone</li> <li>Dial self-provisioning IV number and follow voice prompts</li> </ol>	<ol> <li>Sign in to CM         Administration interface</li> <li>Choose "Migrate         Phone" option in Phone         Configuration page of         old phone</li> <li>Enter phone type         (model &amp; protocol) and         MAC address of the         new phone</li> </ol>
Administrator involvement	Low	Medium	High



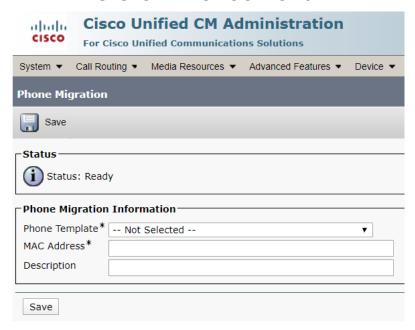
#### CUCM Administration Interface CUCM 11.5 SU8, CUCM 12.5 SU3, CUCM 14



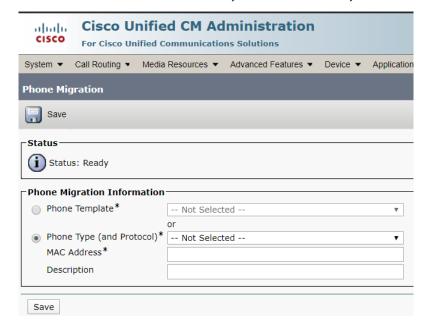


#### CUCM Administration Interface CUCM 11.5 SU8, CUCM 12.5 SU3, CUCM 14

#### Before Enhancement



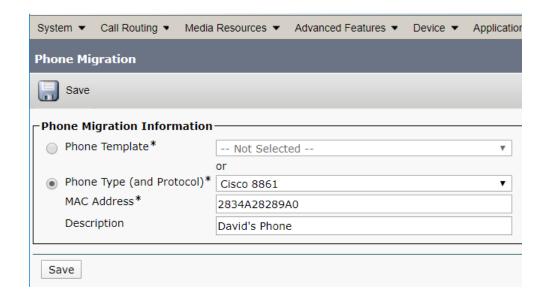
CUCM 11.5 SU8, 12.5 SU3, 14





#### **CUCM Administration Interface** CUCM 11.5 SU8, CUCM 12.5 SU3, CUCM 14

CUCM 11.5 SU8, 12.5 SU3 & 14





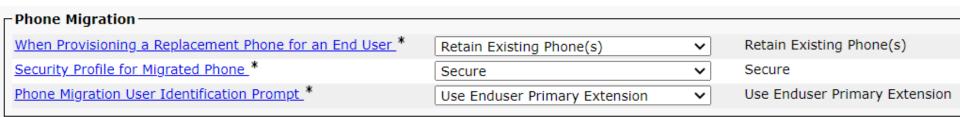
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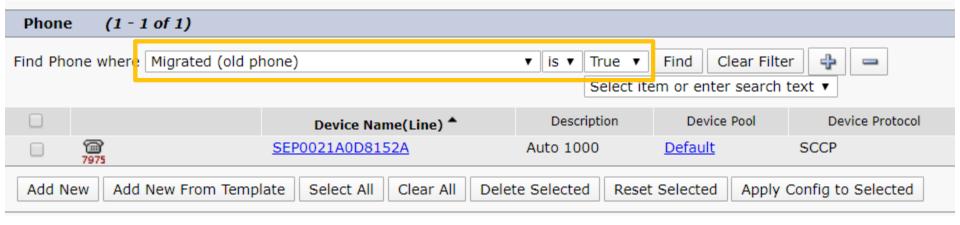
## Enterprise Parameters Configuration

- Two new parameters added
- Default is "Retain Existing Phones"
- You can optionally "Delete the Existing Phone for that End User"
- "Retain existing phones" marks the phone post-migration
- "Secure Profile for Migrated Phone" determines the migrated phones security profile





## Search Migrated phones CUCM 11.5 SU8, CUCM 12.5 SU3, CUCM 14

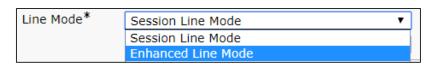




# Simple Phone Refresh + Enhanced Line Mode (Optional)

After phone migration, the administrator can choose one of two methods to move the phone from Session Line Mode to Enhanced Line Mode in CUCM 11.5 SU8.

1. Change the Line Mode parameter on the phone page after migration:



2. Use BAT to enable Enhanced Line Mode

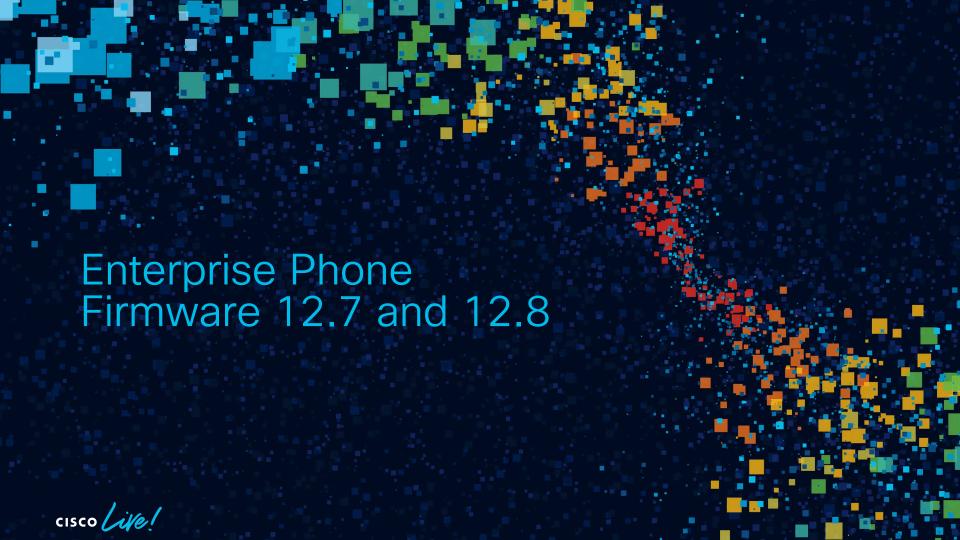




# Multiple Registered Phones http://cs.co/simple-phone-refresh

	Device Status Before Phone Migration	Phone Display During Migration
Scenario 1	Device 1–Registered Device 2–Unregistered	Phone Configuration settings of Device 2 will be migrated.
Scenario 2	Device 1–Registered Device 2–Registered Device 3–Unregistered	Phone Configuration settings of Device 3 will be migrated.
Scenario 3	Device 1–Registered Device 2–Unregistered Device 3–Unregistered	Displays the device list with the following: Description, Phone Model, and MAC Address. Users should choose the device from the list required for phone migration.
Scenario 4	Device 1-Unregistered Device 2-Unregistered Device 3-Unregistered	Displays the device list with the following: Description, Phone Model, and MAC Address. Users should choose the device from the list required for phone migration.
Scenario 5	Device 1—Registered Device 2—Registered Device 3—Registered	Displays the device list with the following: Description, Phone Model, and MAC Address. Users should choose the device from the list required for phone migration.
Scenario 6	More than 3 devices are in Registered or Unregistered states	Error screen: There is more than three device association for <ext.number> extension. Please contact the system administrator.</ext.number>





## New Audio and Video Dual LCD KEMs

New Audio KEM CP-8800-A-KEM= New Video KEM CP-8800-V-KEM= Old Audio/Video KEM
CP-BEKEM=



**USB** charging

100% - space used 4.3 inches

Unused space

4.3 inches 5. width wi

5.5 inches

width

cisco live!

DGTL-BRKUCC-2050

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No angle

for video

phones

No USB

charging

## New in 12.7 Firmware

# Customized KEM Wallpaper

Before 12.7 (old KEM)



Before 12.7 (new KEM)



After 12.7 (old KEM)



After 12.7 (new KEM)



# Customized KEM Wallpaper

Before 12.7 (old KEM)



After 12.7 (old KEM)



- Upload pictures to CUCM TFTP server, the pictures resolution should be fit phone or KEM screen.
   The pictures need to be the same name in all directories. Example: cisco1.png
  - Desktops/800x480x24 --- for main phone screen wallpaper
  - Desktops/320x480x24 --- for the new 8800-A/V-KEM wallpaper
  - Desktops/272x480x24 --- for BEKEM wallpaper
- 2. Restart TFTP. On a new Common Phone Profile, uncheck "Enable End User Access to Phone Background Image Setting. For "Background Image" put wallpaper (i.e., cisco1.png)
- 3. Apply Common Phone Profile to the phone.



# Wallpaper Whitepaper Update

# Go to <a href="http://cs.co/custom-wallpaper">http://cs.co/custom-wallpaper</a>

### or

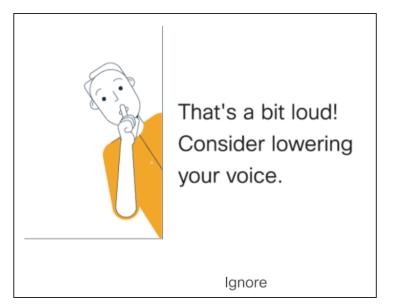
 https://www.cisco.com/c/dam/en/us/products/collateral/collaborationendpoints/unified-ip-phone-8800-series/white-paper-c11-740036.pdf

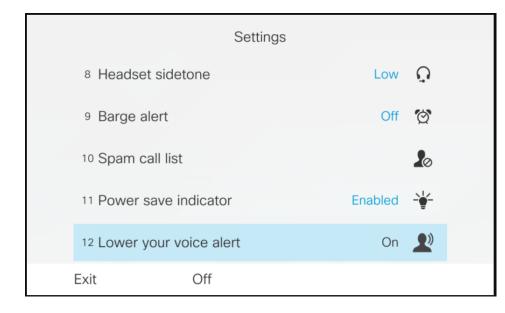




## Lower Your Voice

- Loud voice will trigger alert to lower your voice
- Useful in open office environments
- Works with handset or headset

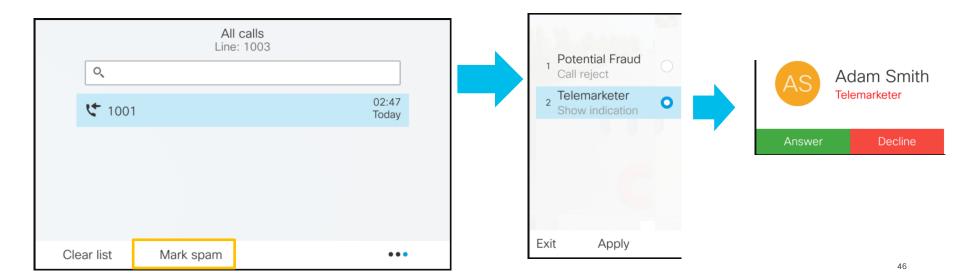






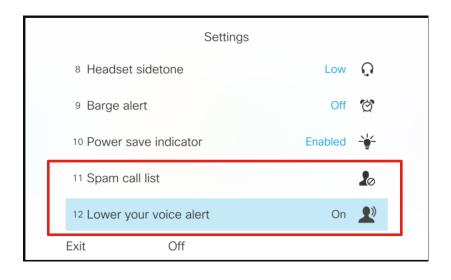
# Spam Detection

- Allows end user to classify calls that are potential fraud or telemarketing using a softkey in call history or in a call
- Choosing "Potential Fraud" blocks the calls
- Choosing "Telemarketer" allows future calls but flags it

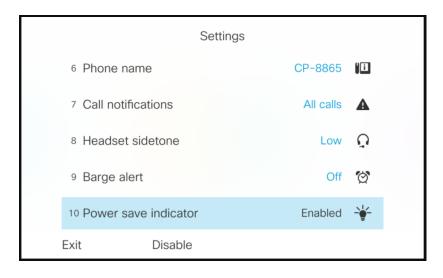


# Lower Your Voice and Spam Call

## 12.7 Firmware



## After Upgrade to 12.8



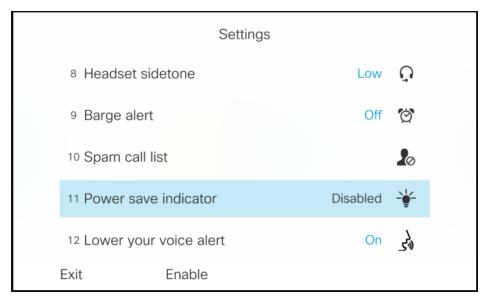


# Disable Circular Light

#### Use cases:

- Hospitality
- Boats
- Environments requiring darkness





## 8821-EX

- Hazardous Locations certification (METLABS listing: E114610)
  - ANSI/ISA 12.12.01 & CAN/CSA C22.2 No. 213 Class I & II, Division 2 and Class III, Division 1 & 2
  - Atmospheres Explosibles (ATEX) Zone 2 \*
- IP67 rated for dust and water protection in ordinary locations
- MIL-STD-810G Standard for durability
- HD voice quality and full-duplex speakerphone
- Wi-Fi IEEE 802.11a/b/g/n/ac
- Fast Secure Roaming (11r/FT, CCKM)
- SHA-256 certificate signature and 4096 bit key support
- Easy Provisioning (Dock + USB Dongle, CUCM WLAN Profiles, SCEP)
- Bluetooth 3.0
- XSI Application support





# What are Multiplatform Phones?

- Open SIP Phones
- Connecting to:
  - Cisco Webex Calling
  - Cisco Broadsoft BroadCloud
  - Cisco Broadsoft BroadWorks
  - Other Cisco approved Unified Communications-as-a-Service (UCaaS) platforms

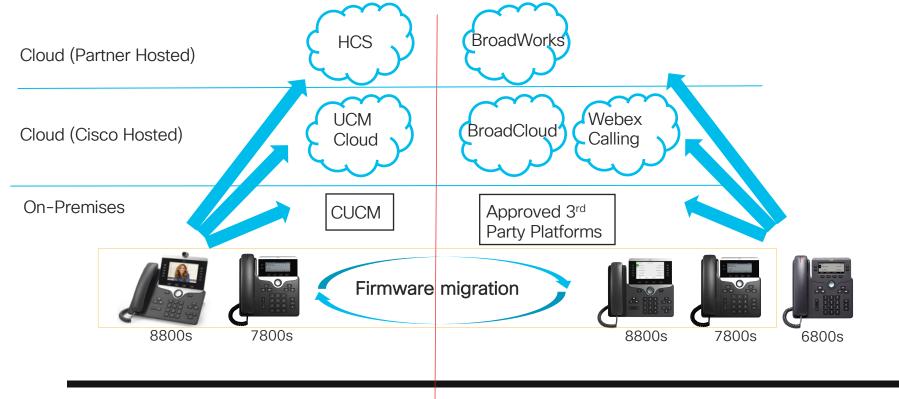
Successors to the now, End of Sale SPA phones, MPP firmware has many of the same traits as the SPA phones.

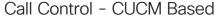
- Developed for the SMB market
- Managed by Service Providers
- Interoperability tested with third-party SIP-based systems
- Same rock solid quality as the On Premises Cisco IP Phones



#### **Evolution of MPP**

# A Brief History of Cisco IP Phones

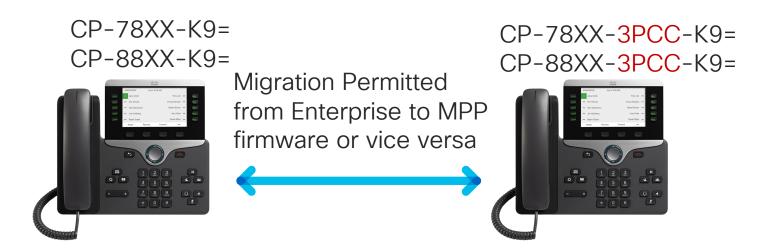




Call Control - Open SIP Based



# Firmware Migration Options for 7800/8800



- Existing on-premises phone migration permitted
- Buying new phones and immediately migrating is not permitted
- Costs waived under an eligible Flex plan

# Migration process overview http://upgrade.cisco.com



#### Check eligibility

- Check device eligibility (see Table 1 in the <u>overview</u> <u>section</u>)
- Check your subscription for eligible Webex Calling SKU (see Table 4 in the <u>FAQ</u> <u>section</u>)



#### Order licenses

- Appendix A: How to place an order
- Appendix B: Generate licenses for your devices



#### Set up and prepare

- Upload licenses to the <u>Cloud Upgrader</u> tool
- Add devices to the Webex® Calling portal
- [Optional] Upgrade phones to the latest firmware if registered to UCM
- [Optional] <u>Set up</u>
   <u>DHCP options</u> if staging the phones



#### Migrate phones

- Upgrade phones by following instructions on upgrade.cisco.com
- Upgrade via UCM or use a setup with DHCP options
- After upgrading to migration firmware, the rest is automatic



## Resources



- Phone Firmware Migration detailed guide
- Cloud converter <a href="https://upgrade.cisco.com">https://upgrade.cisco.com</a>
- Do it yourself
  - Enterprise to MPP conversion guide
  - Phone Firmware Migration Bulk Provisioning guide

# The Cisco lineup of Multiplatform Phones

## **Cloud Only**

#### **CUCM or Cloud**

#### Cisco 6800 Series



- New family of entry-level phones
- The only IP phones exclusive to multiplatform
- Enterprise-grade quality with a user experience similar to the 7800/8800 Series

#### Cisco 7800 Series



- Ideal for lightly-to highly-active voice users
- High-quality wideband audio
- Easy-to-use
- Backlit grey scale displays
- Speakerphone on all models

#### Cisco 8800 Series



- Ideal for knowledge and administrative workers, managers, and executives
- Color display and Bluetooth
- Cisco Intelligent Proximity for smartphone integration
- Optional key expansion modules



## Cisco IP Phones 6800 Series







	6821	6841	6851
Screen	240 x 120 Backlit Mono (2.5")	396 x 162 Backlit Mono (3.5")	396 x 162 Backlit Mono (3.5")
Ethernet Switch	10/100	10/100/1000	10/100/1000
Programmable Line Keys	2	4	4
Programmable Soft Keys	2	4	4
Headset Port			
EHS Support (AUX)			
Full Duplex Speakerphone			
Wideband Audio		Handset Sold Separately	
POE Class	2		2
Optional KEM			1
Footstand	Single, upright position	Single, upright position	Single, upright position
Wall Mount			



## Cisco IP Phone 6861

- Made for install challenging environments – Asbestos, Stone Walls, etc.
- 4-Line IP Phone
- Dual-Band Embedded Wi-Fi support 802.11a/b/g/n
- AP Auto Scan
- Full spectrum of CODEC support including Opus



## Cisco IP Phone 6871

- 3.5" color LCD screen (480x272)
- 6 Programmable Line Keys
- Electronic Hook Switch (AUX)
- 1 x USB-A port for Headsets
- POE
- Gigabit Switch





## Cisco IP DECT 6825 Handset and 210 Base Station

- Excellent Indoor / Outdoor Range
- Great Roaming / Mobility
- Secure Radio Communications
- Minimize installation costs, reduce time to dial tone
- Bluetooth and 3.5mm Headset jack
- Built-in Belt Clip





# Comparing Multi-Cell and Single Cell DECT

Multi-Cell System		Single Cell System
Buildings, Retail, Factories, Etc.	Ideal For	Small Office, Home Office
Start Small, Pay As You Grow	Investment Protection	Start Small, Replace Infrastructure
Slight Premium	Cost	Entry Level Pricing
Up to 1,000 Phones	Capacity	8-10 Phones Max (Typically)
Add Basestations, extend Range and Capacity	Expansion	Add Repeaters to Extend Range, Repeaters limit capacity
Seamless handoff between basestations in the system	Mobility, Roaming	Roam within range of basestation and repeaters

## MPP 6800 series - DECT Deployment Guide

https://www.cisco.com/c/dam/en/us/td/docs/voice\_ip\_comm/cuipph/MPP/6800-DECT/deployment/CiscoDECT6800DeploymentGuide.pdf

# DBS-110 Single Cell Base, Repeater, & 6823 IP DECT Phone



### Cisco DBS-110

- Single Cell Basestation and Repeater
- 30 SIP registrations
- 10 concurrent calls
- Differentiated by icon:











## Cisco 6823 IP DECT phone

Color Display

Wideband Audio Support

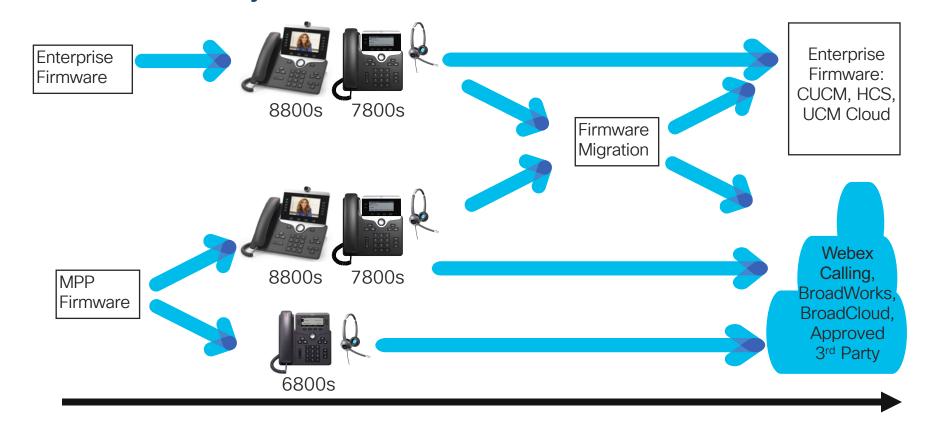
Same battery, and talk time as Cisco 6825 IP DECT phone

Can be used with our multi-cell or single cell basestations and repeaters



## A Brief History of Cisco IP Phones

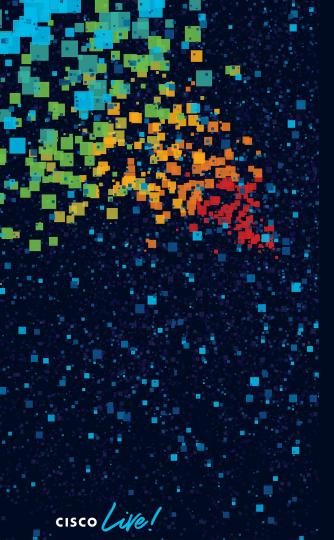
### **Evolution of MPP**











# Agenda

Part 1 - IP Phone Update

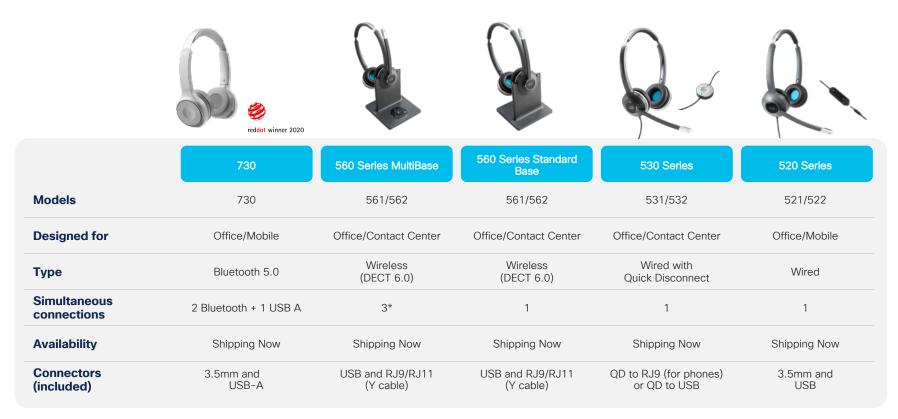
Part 2 - Headset Update

- Cisco Headset 500 Series
- Extension Mobility Headset Login
- Headset Management
- Cisco Headset 730 Series

Part 3 - Conference Phone Update

Webex Room Phone

# Cisco Headsets model comparison



<sup>\*1</sup> Bluetooth Device + 2 Wired Devices (2 USB or 1 USB + 1 RJ9/RJ11)



## End User Headset Customization and Cisco Phones

Headset Customization Availab	ility 8851/61/65	8811/41/45	78xx
USB Cable (52x, 53x, 56x, 730	)) Yes*	-	-
Y-cable (56x)	Yes	Yes	Yes

\*500 series requires phone firmware 12.5(1)SR3 or later



Test microphone

#### Gain:

Adjust the microphone gain

#### Equalizer:

Adjust audio tune/bass

#### Side Tone:

Adjust sound from microphone





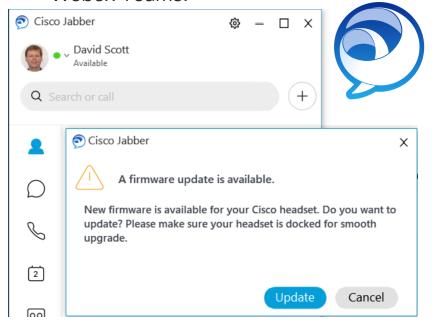
<sup>\*730</sup> series requires phone firmware 12.7 or later

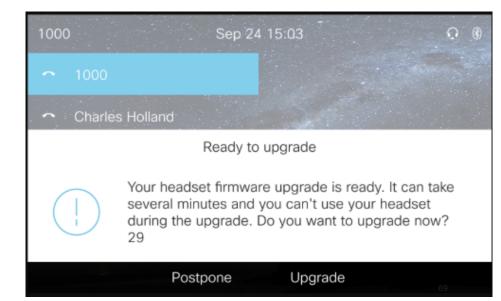
# CUCM Upgrade Support via Phones or Jabber

- Headset firmware can be pushed to the 500 Series from any version of CUCM.
- Headset management for 500 series or the 730 requires CUCM 11.5 SU7+ or CUCM 12.5 SU1+. The 730 requires these versions for firmware upgrades as well or you can use a smartphone app to upgrade the Cisco Headset 730.

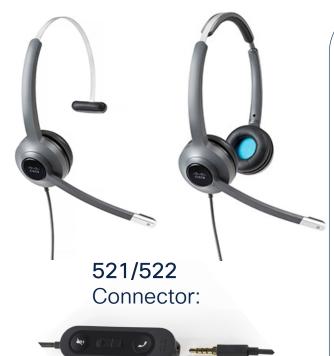


 You upgrade the headset using 7800/8800 running 12.5(1)SR3+, Jabber 12.6MR1+, or Webex Teams.





# Cisco Headset 521/522 and 531/532



521/522 includes a 3.5mm connector and includes an in-line USB adapter

- Wired Headsets
- Single and Dual Ear Piece Models
- Busy Indication
  - · On-ear and on controller
- Powerful Noise Reduction
  - Excellent in open offices

#### **USB** indicator

Incoming Call	Blinking Green	Ç
Active Call	Steady Green	Ç
Muted Call	Steady Green	Ç
	Steady Red	•



531/532 Connector Options:





USB

for non-USB Cisco Phones

## Cisco Headset 561/562 with Standard Base or Multibase



Standard Base Connectivity: 1 USB or RJ9/11 cable

CP-HS-WL-561-S-EU= CP-HS-WL-562-S-EU= Frequency Band: Europe, UK, Asia,

South Africa, HK, Australia

#### Wireless

- Exceptional Clarity
- Extended Range (DECT 6.0)
- Best-in-Class Audio
  - Adaptive for voice and music
- Easy Enterprise Deployment
  - User-based Policy
  - Automatic Software Upgrades
  - Native Integration for Plug & Play
- Robust On-Ear Controls
  - Advanced call handling, mute, volume
- Security
  - 128 Bit Step C (Highest DECT Standard)
- Easy to Order
  - · Cisco cabling included



### Multibase Connectivity:

1 USB or RJ9/11 cable

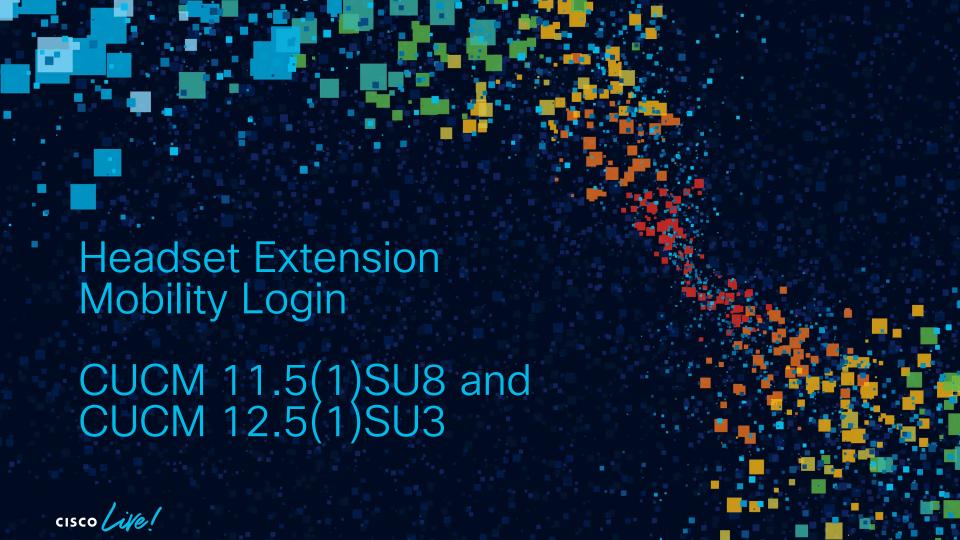
1 USB to laptop or PC

1 Bluetooth connection

CP-HS-WL-561-M-EU= CP-HS-WL-562-M-EU=

Frequency Band: Europe, UK, Asia, South Africa, HK, Australia





# Extension Mobility login simplification with headsets GA release: CUCM 11.5 SU8

Use the user identity associated to a headset to simplify Extension Mobility login to a CUCM / UCM Cloud registered phone



#### New Extension Mobility login option:

- Plug headset into an Extension Mobility enabled phone. Identity in the headset will perform an Extension Mobility login to the phone.
- Security conscious customers can enforce PIN entry as well.
- Enables headset for hot-desking in Open Workplace



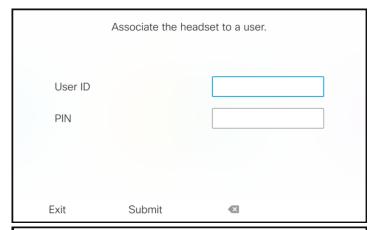


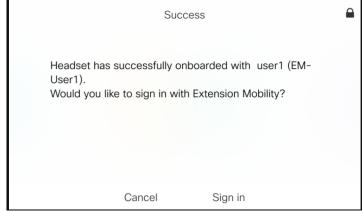


# Headset Extension Mobility Login CUCM 12.5 SU3 & CUCM 11.5 SU8

#### How it works:

- Connect headset to trigger EM login
- Disconnect headset to trigger EM logout
- Headset can be associated to the end user in three ways
  - End user can associate themselves to a headset for EM via a screen pop up
  - End user can associate themselves to a headset for EM via the phone menu
  - The admin can associate the end user to a headset serial number in CUCM



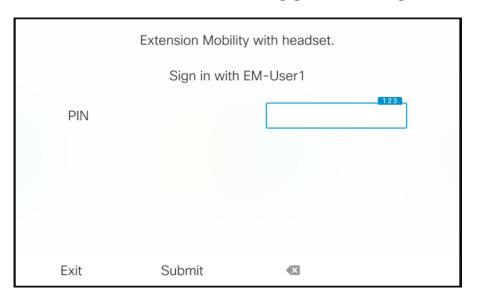




# Headset Extension Mobility Login CUCM 12.5 SU3 & CUCM 11.5 SU8

Once onboarded ...

#### Connect headset to trigger EM login



#### Disconnect headset to trigger EM logout

Extension Mobilit	y with headset.
Sign out EM-User1 from this You will be signed out autom	
Cancel	Sign out

# Headset Extension Mobility Login CUCM 12.5 SU3 & CUCM 11.5 SU8

### Supported phone/headset list with phone firmware 12.8 or higher:

- 1. 8851/51NR/61/65 with headset 521/522/531/532 via USB
- 2. 7821/41/61 and 8811/41/45 with headset 561/562 via Y cable (RJ9/RJ11)
- 3. 8851/51NR/61/65 with headset 561/562 via USB or Y cable (RJ9/RJ11)
- 4. 8851/51NR/61/65 with headset 730 via USB





#### 1.5 Firmware and later\*

# Cisco Headset 560 Series Conferencing

- Allow a maximum of 3 guest headsets to pair with one dock (4 total)
- All headsets can talk to each other anytime, even if there is no active call
- Guest headsets join the call muted. They can unmute themselves if desired.

#### Connect:

- 1. Dock Guest headset in primary dock
- 2. Primary headset receives tone indication
- 3. Primary headset confirms with call button

#### Disconnect:

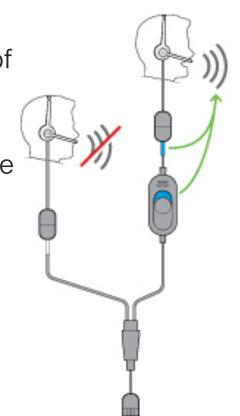
Place the primary headset into primary dock

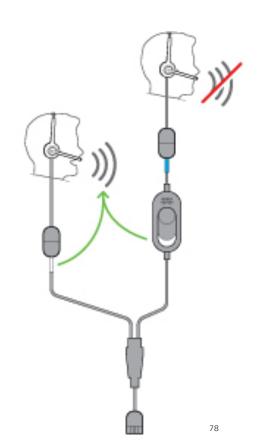


\*Enabled by default in HS2.1

# Cisco Headset 530 Series Y Training Cable

- You can connect a second headset for the purposes of training
- A switch on the cable allows only one microphone to be active at a time
- The color on each arm of the cable corresponds to the colors on the switch





# Multibase: Answer Call on Another Source

In 1-5-1-15, if there is incoming call on inactive source, the user will be not able to answer it via headset button

In firmware 2.0, the user can answer an incoming call on the inactive source via the headset button

#### Example:

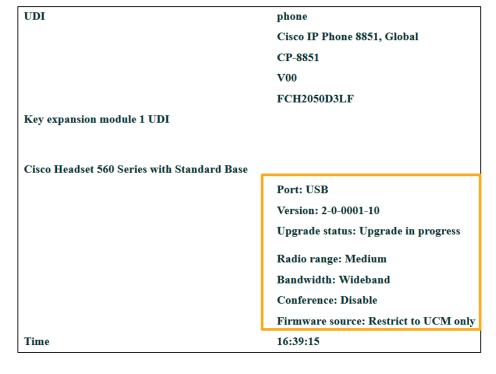
- Cisco IP phone connects via USB-A
- PC connects via micro-USB
- While listening to music on PC, there is an incoming call on the Cisco IP phone
- Pressing call button on headset will answer the incoming call and auto-select the source USB-A



# Show Admin Settings on Phone Web Page

The administrator can now see admin settings on phone webpage:

- Upgrade Status
- Upgrade Source
- Port: How the headset connects
- Version: Headset firmware
- Radio range: DECT Radio Range
- Bandwidth: Narrowband or Wideband
- Bluetooth: Enabled or Disabled
- Conference: Enabled or Disabled

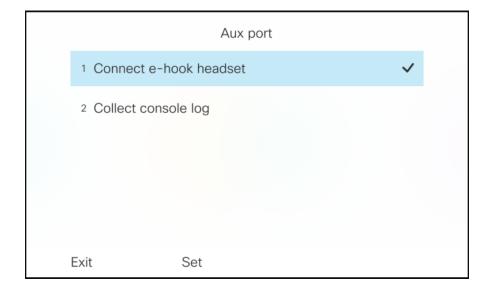


DGTL-BRKUCC-2050

## E-Hook By Default for 7800/8800

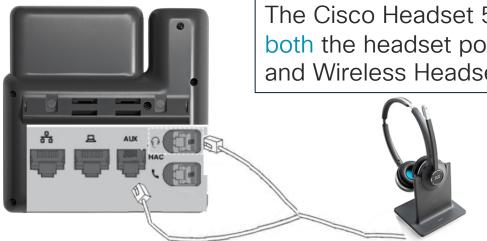
Settings -> Aux Port -> Connect e-hook headset

Please select a menu item				
<sup>1</sup> Network setup	>			
<sup>2</sup> Security setup	>			
3 Status	>			
4 Aux port	>			
5 Reset settings	>			
Exit				





#### Wireless Headset Hookswitch Control and Y-cable



The Cisco Headset 560 Y-Cable must be plugged into both the headset port and the AUX port on the phone and Wireless Headset Hookswitch Control Enabled

RESOLVED IN 12.7
FIRMWARE. PHONE WILL
AUTOMATICALLY ENABLE
THIS PARAMETER

Wireless Headset Hookswitch Control\*

Enabled



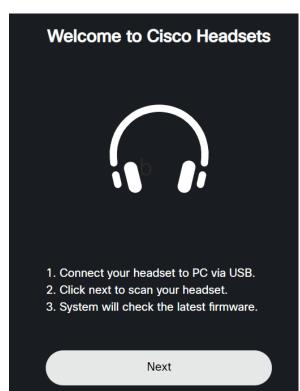
Important! It is required to enable Wireless Headset Hookswitch Control parameter on CUCM in order for the Y-cable to work on the 78xx or 8811/41/45

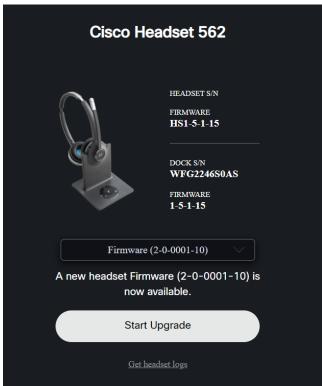


# Cisco Headset Web Tool (beta)

Display Headset Info, Firmware Upgrade and Log Collection

- This is a web browser
  headset tool for customers
  unable to upgrade the
  headsets due to lack of
  CUCM, Cisco IP Phones,
  Jabber, or Webex Teams
- Available for Cisco 560 and 730 series headsets
- Supported browser:
  - Google Chrome develop version latest(84+)
  - Enable development flag (chrome://flags/#enable-experimental-webplatform-features).







# Multiple Application Call Control Handling

 Before: if you are running more than one soft phone application (such as Cisco Jabber, Webex Teams, WebEx Meeting App, Microsoft Skype for Business, Microsoft Teams) on your PC connected with Cisco headsets, you may experience unexpected call control behavior due to conflicts between applications.

• What's new: With the latest Cisco Jabber / Webex Teams / Cisco Headset release, we will provide a better Cisco on Cisco user experience when multiple applications are used at the same time. Cisco is the only headset vendor who is able to support multiple calls between applications.



# Typical User Scenarios

- Multiple Incoming Calls
  - The user has both Cisco Jabber and Webex Teams running on the PC.
  - When there are incoming calls from both Cisco Jabber and Webex Teams, the user can press the call button on headset to answer the call, and the incoming call will follow a first come first serve basis.

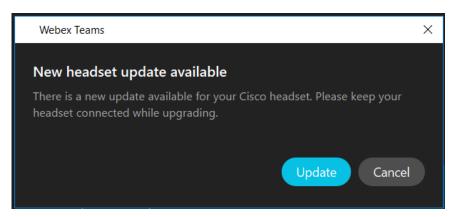
- Multiple Calls Between Apps
  - The user has both Cisco Jabber and Webex Teams running on the PC.
  - There is an active call on Cisco Jabber. The headset is controlling Jabber.
  - The user answers or initiates a new call on Webex Teams from the PC.
     The headset switches call control from Jabber to Webex Teams.
  - After Webex Teams call ends, the headset switches its call control back to Cisco Jabber.



## Webex Teams Integration with Cisco Headset

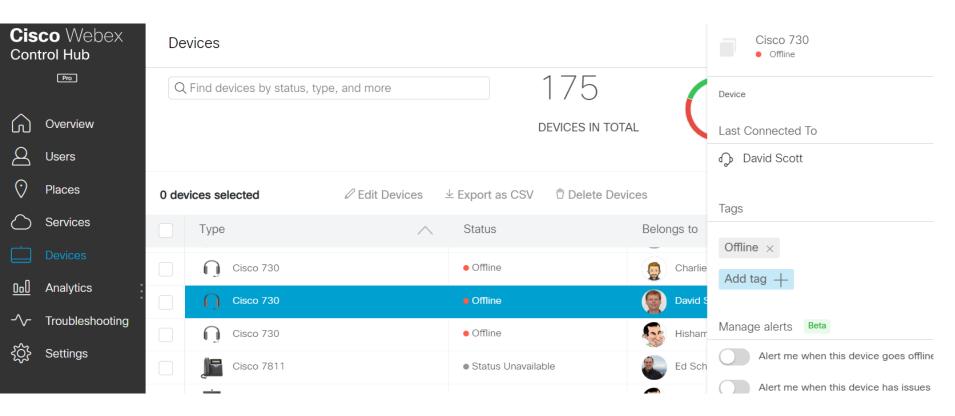
Headset upgrades can now be pushed from Webex Teams

- 1. There is no CUCM dependency
- 2. Webex Teams client detects existing model and firmware
- 3. Webex Teams client upgrades to newer firmware
- 4. Will work with Cisco Headset 500 and 730 series





# Headset Inventory in Webex Control Hub

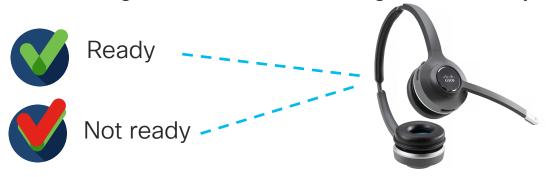




## Headset Integration with Finesse

### Change Finesse Ready/Not-Ready status using only the Cisco Headset

- Between calls, agents can toggle their status to Ready for a call or Not Ready by clicking the headset's Mute button.
- 2. During calls, the Mute button toggles the status to Mute or Unmute, maintaining current behavior during calls.
- 3. The headset base station LED displays the agent status (for example, Ready, Not ready) so that other agents know whether the agent is active by looking at the LED.



#### Requirements

- 560 Headset 2.1 Firmware
- Cisco 8800 12.8 Firmware
- CUCM 11.5 SU8

https://www.cisco.com/c/dam/en/us/td/docs/voice\_ip\_comm/cucm/whitePaper/CUCM\_Headsets\_for\_ContactCenter\_WP.pdf





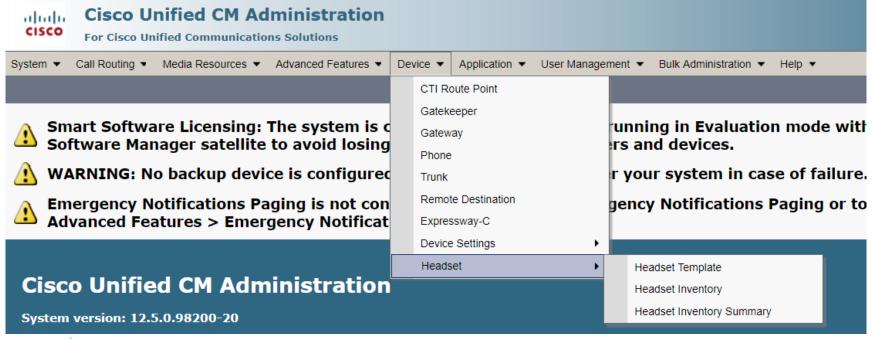
## CUCM 12.5 SU1+ and 11.5 SU7+ Headset Support

- Inventory Management: CUCM will provide reports based on headset model, connection status, firmware releases, connections, and more
- Remote Configuration: CUCM will be able to control headset settings, including wireless power range, wideband/narrow band settings, firmware version, Bluetooth on/off, and more (along with templates to help guide administrators)
- Analytics and Diagnostics: CUCM call records (CMRs) will be enhanced with additional metrics from headsets, such as RSSI (wireless signal strength), frame errors, connection drop reason, beacon moves, audio settings, DECT bandwidth, and more
- Remote Log collection: The CUCM user interface and the Real Time Management Tool (RTMT) will be able to trigger log collection, including headset logs without any user involvement.
- Firmware upgrade: CUCM can push new firmware to headsets using Jabber and IP phones, without the need for extra headset management software. This capability is available today. With CUCM 12.5, administrators will be able to control firmware versions from a configuration template.



### Headset Management CUCM 12.5 SU1+ & CUCM 11.5 SU7+

## Live demo here — cs.co/headset-management

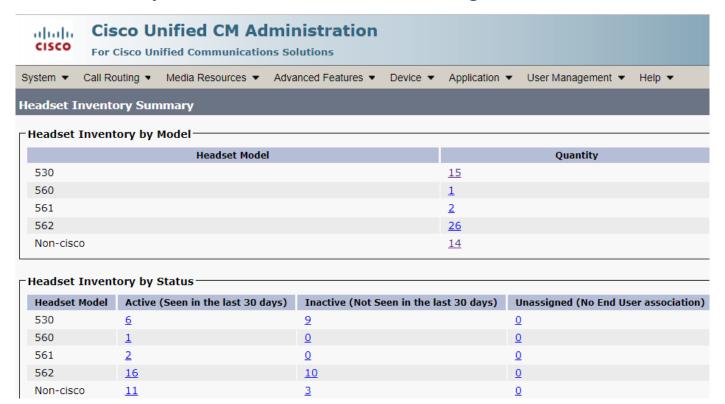




## Headset Inventory

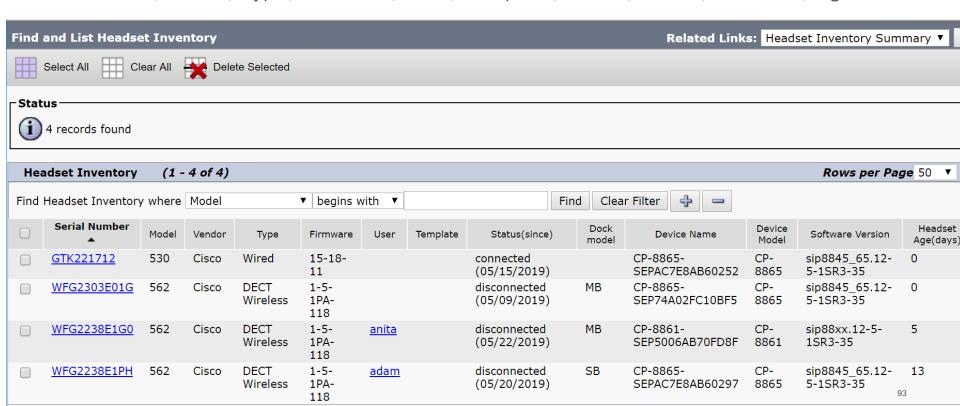
Headset Inventory by Model and Quantity

Headset Inventory Status: Active, Inactive, Unassigned



## Headset Inventory Detailed

Serial Number, Model, Type, Firmware, User, Template, Status, Phone, Software, Age



### Cisco Headset Service

- There is a new "Cisco Headset Service" in CUCM 12.5 SU1+ and CUCM 11.5 SU7+
- This service is required to be activated on both the publisher and subscribers in the cluster

Cisco Intercluster Lookup Service

Cisco Location Bandwidth Manager

Cisco Directory Number Alias Sync

Cisco Directory Number Alias Lookup

Cisco Headset Service

Cisco Device Activation Service

Cisco Dialed Number Analyzer Server

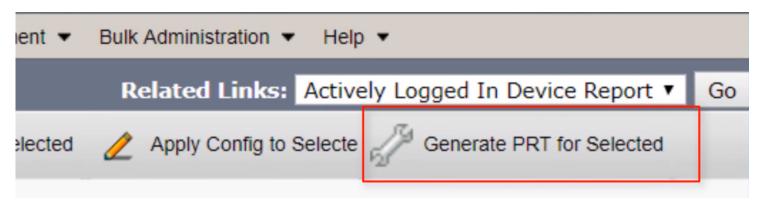
Cisco Dialed Number Analyzer

Cisco Tftp



## Troubleshooting

- CUCM 12.5 SU1+ or 11.5 SU7+ can now remotely generate a Problem Report for 7800/8800 devices you select
- This greatly facilitates troubleshooting a headset attached to the phone
- The 7800/8800 phone must be running 12.5(1)SR3 or higher





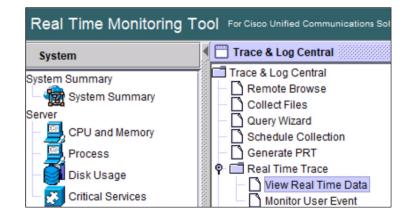
## Diagnostics and Alarms

Headset will send related call information to Call Management Records

#### Data to be collected from headset:

Wireless Info - every call

- RSSI relative signal strength (average, min, max)
- Frame Errors (total for the call)
- Call/Connection drop reason (N/A if not applicable)
- Beacon moves
- DECT bandwidth





# Headset Remote Configuration Minimum Requirements

	Pre-CUCM 12.5 SU1 or 11.5 SU7 (Global Policy)*	CUCM 12.5 SU1 (Per User Group Policy)
7800/8800 Phone	12.5(1)SR3	12.5(1)SR3
Jabber*	12.6MR+	12.6MR+

- Speaker Volume
- Microphone Gain
- Sidetone
- Equalizer (Tune)

- Audio bandwidth
- Bluetooth
- DECT Radio range
- Firmware Upgrade

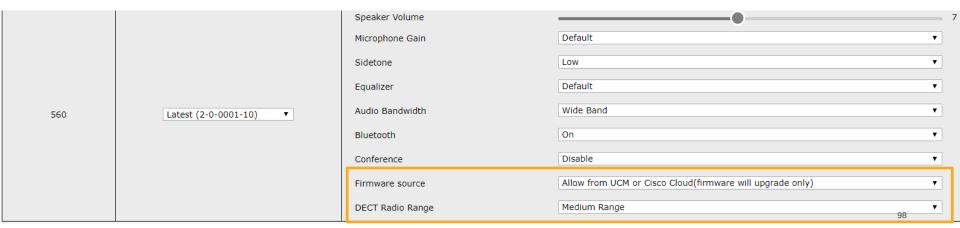
\*Jabber 12.0 supports Cisco headsets 12.5 adds software upgrade

12.6 will support config management.



# Cisco Headset 2.0 Template Update

- New Firmware Source added
  - Allow from UCM or Cisco Cloud (firmware will upgrade only)
  - Restrict to UCM only (firmware may upgrade or downgrade)
- Headset Density & Power range enhancements
  - Medium (default) balanced mode between user density and mobility
  - Auto sparse workspace favoring roaming over user density
  - Short densely populated space favoring user density over roaming



## Cisco Headset DECT Whitepaper

DECT Whitepaper

# Cs.co/dect

YouTube video walkthrough of whitepaper

https://youtu.be/7uGqDsRHWMI

How to Deploy
DECT at Work for the
Cisco Headset 560 Series

DECT deployment options for the Cisco Headset 560 Series

Setting Name	Option	Description
DECT radio range	Medium range (available in November 2019)	Medium transmitting power
	Short range	Low transmitting power
	Auto range	High transmitting power
Audio bandwidth	Narrowband	Lower bit rate
	Wideband	Higher bit rate



## Cisco Headset 730: 360° view





### Cisco Headset 730

Premium Bluetooth Headset for business, personal and the journey between

- Innovative beam forming, Boomless technology enabling crystal clear calling
- Your Focus and Collaboration zones, one touch away
- Deep Cisco integration to extend the cognitive collaborative experience
- First Bluetooth headset with enterprise grade security
- Bluetooth, USB, and 3.5 mm connectivity options
- On-prem and Cloud solutions Management and Metrics



### Cisco Headset 730



HS-WL-730-BUNA-P



HS-WL-730-BUNA-C

- Sleek microphone design
- On-ear buttons for easy call and music control
- // Integrated presence LED light

Manual "Do not disturb"

Call presence

- Replaceable leather ear cups
- 2 colors options:

Platinum

**Carbon Black** 

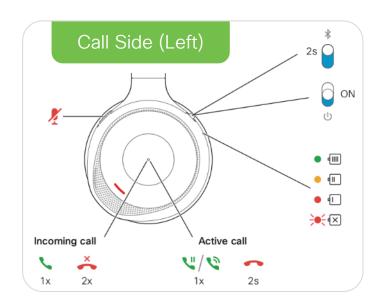
### Premium audio and voice

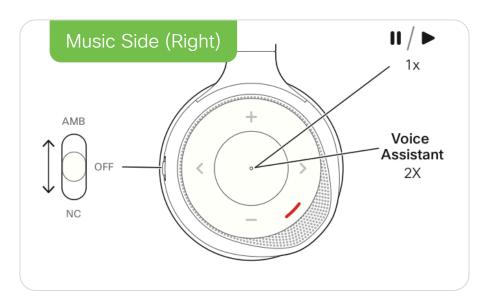


- Clear Voice technology: Cisco's voice honing, boom less design coupled with adaptive noise cancellation, background noise suppression, and side tone enabling crystal clear audio. Anytime.
- Switch between noise cancelling modes:
  - Noise cancellation (NC) block out environment noise for optimal calling and music experience
  - Ambient (AMB) ability to hear conversations in user's environment
- Premium codecs and 40 mm speakers for exceptional calling and music experience
- Adjustable equalizer settings
- Voice activated Al: Access Google Assistant, Siri, or Cortana through your headset



# Ear controls and battery





Battery

Talk time with ANC **18+ hours** 

Music/listen time with ANC

25+ hours

**Charging Time** 

150 mins (Full charge)

10 mins (Quick charge for 1-hour talk time)



## Power/Bluetooth

- The Power/Bluetooth switch is located on the back of the left ear cup.
- To **turn on**, slide the Power/Bluetooth switch up and let go. A voice notification will state remaining battery life.
- The power LED light also signals the **battery life**.
- While wearing the headset, slide the switch up and let go to hear the remaining battery life.
- To **turn off,** slide the power/Bluetooth switch down.

# Pairing Mode

- Slide the Power/Bluetooth switch up.
- Hold for 2 seconds until you hear "Pairing mode, look for Cisco Headset 730 in your list of Bluetooth devices", and the Bluetooth LED will blink.
- The headset is now in pairing mode and is discoverable.
  - Once paired, you will hear "Connected."



# Noise Cancelling Switch

The Noise Cancelling switch is located on the back of the right ear cup.

- Slide the switch down to turn on noise cancellation (NC)
- Slide the switch up to turn on ambient (AMB)
- Slide the switch in the center to turn off both AMB and NC





#### Reference

### Call control side - left

- The call control side is located on the face of the left ear cup.
- Press the call button **once** to answer an incoming call.
- Press the call button **once** to hold an active call.
- Press the call button **twice** to reject an incoming call.
- Press and hold the call button for two second to end a call.





#### Reference

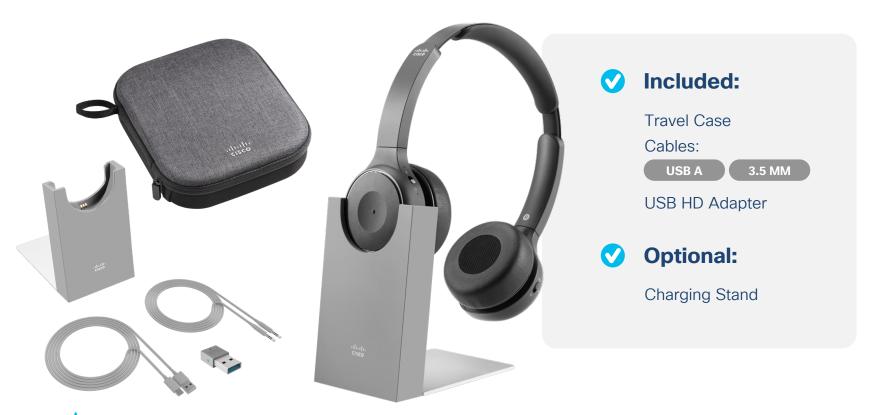
# Music control side - right

- The music control side is located on the face of the right ear cup.
- Press < or > to skip a track on your music.
- Press + or to adjust your music or call volume.
- Press the center button to pause or resume your music playback.
- If you have Auto Pause/Play enabled in the Cisco Headsets app, it will automatically pause music playback when taking off the headset or resume music playback when putting on headset.





### Cisco Headset 730 accessories





### Headset support minimum requirements

	Connection	CUCM	8800 Support USB or Bluetooth	8800 Phone Firmware	Jabber	DX and Desk Pro
730	Bluetooth 5.0, USB A, and 3.5mm	All versions for basic audio. Upgrades and management require 11.5 SU7+ or 12.5 SU1+	8851/61/65	12.7	12.7*	CE or RoomOS



Simultaneous Connections: 2 Bluetooth + 1 USB





CUCM 11.5(1)SU7+ and 12.5(1)SU1+ Firmware Updates and Remote Configuration



# 8800 phone support for Cisco Headset 730 Series

Users can customize their headset from the phone screen. The 730 USB cable can connect to the 8851/61/65 for:

- Charging the headset
- Firmware upgrade\*
- Customization\*
- Inventory management\*

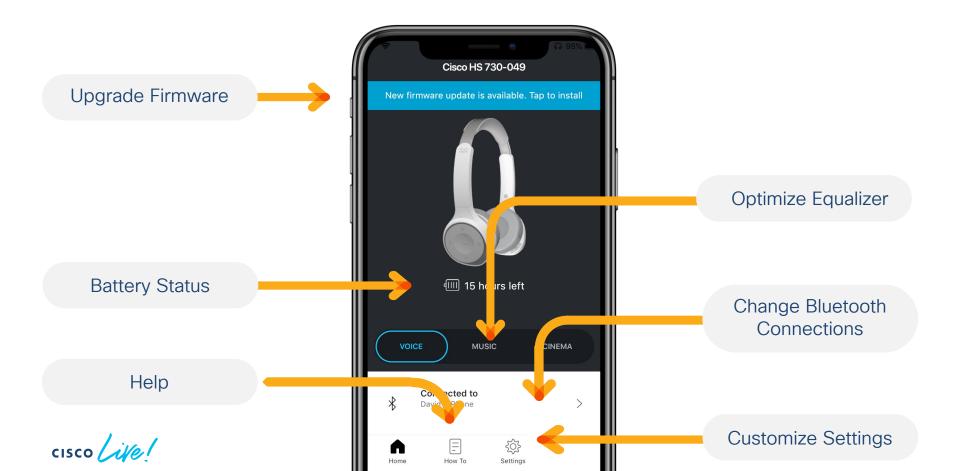
<sup>\*</sup> Requires CUCM 12.5 SU1 or 11.5 SU7 or later





Connection Type	8845	8851	8861	8865
USB (Adaptor or USB-C cable)	Not supported	Supported	Supported	Supported
Bluetooth (pairing with phone)	Supported	Supported	Supported	Supported

# Cisco Headset app for iPhone and Android







### Today's huddle spaces restrict collaboration



only have local sharing





have audio and video

The result?

Departments buy their own

Low quality, weak security products

No standard across the enterprise



Source: Wainhouse - 2017

### Cisco Webex Room Phone

# Superior Audio quality

360-degree HD audio with echo cancellation and noise reduction

#### Smart meetings

Flexible sharing – wireless, wired and guest, Active speaker and participant list



#### Easy Meeting join

Room calendar, One button to push and zero touch proximity with Webex apps

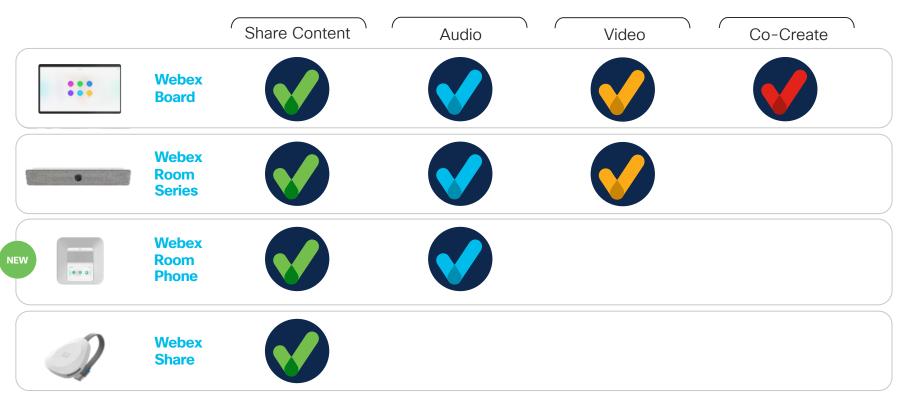
#### Simple Management

Control Hub to provision and monitor usage

Using the power of the Webex platform and a simple, familiar interface.



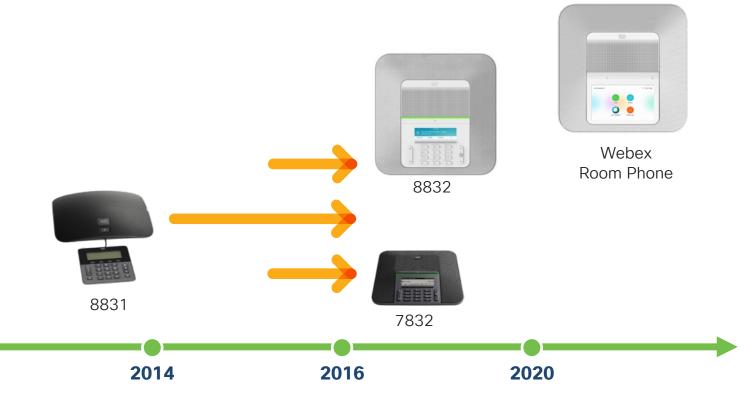
# Cisco huddle & meeting space devices



SIMPLE | CONNECTED | SCALABLE | COGNITIVE



# Evolution of conference phone



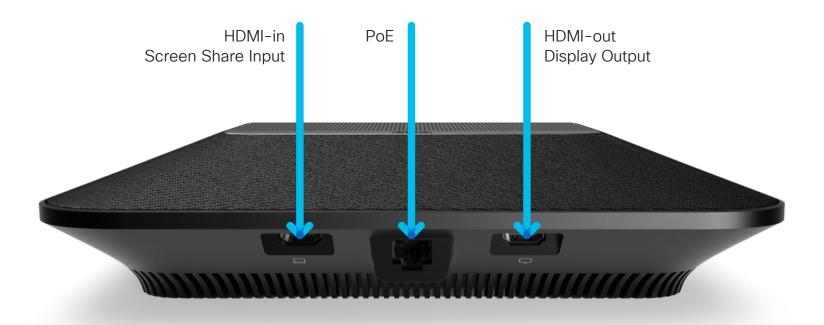


### Cisco Webex Room Phone



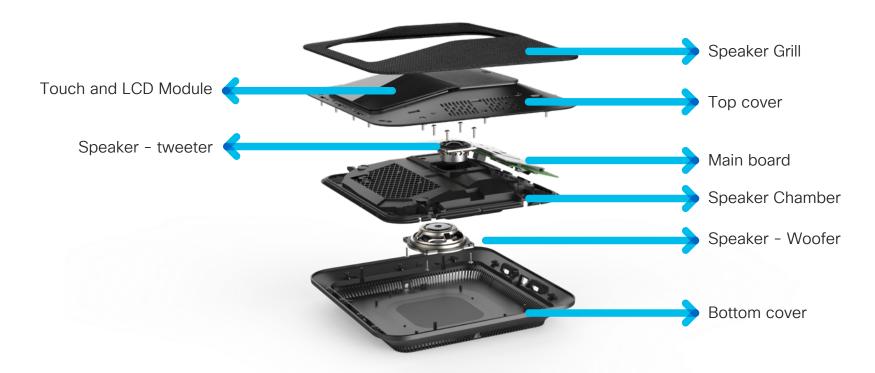


### Room Phone - backplane connectors





### Room Phone - internals





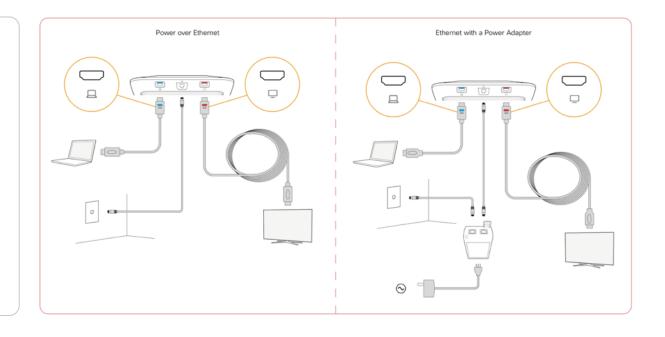
### Room Phone - installation

#### **Network and Power options:**

POE (Class 3 devices) Connect network cable between RJ45 port on device and wall port

#### Non-POF

- Connect network cable between RJ45 port on device and POE injector port
- Connect POE injector to power source and to wall port





# Great audio for huddle and beyond

#### **Studio monitor full-range speaker**

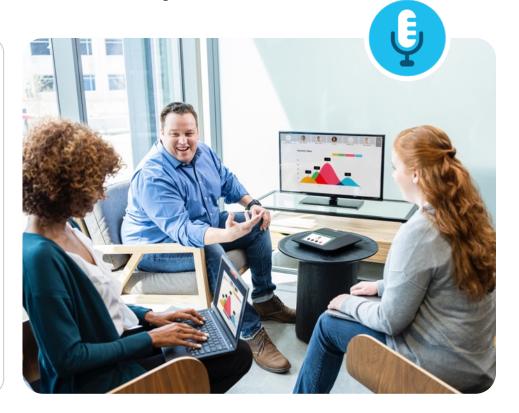
dynamic, detailed sound with low distortion and low-frequency clarity provides exceptionally clear and natural audio quality

#### **Reduce distractions**

Acoustic echo cancellation and noise reduction provides Full-duplex audio without echo at 50 cm even in long reverberation (around 1sec) meeting rooms

Dynamic Noise Reduction reduces background noise without impacting audio quality

Comfort Noise Generation Improves subjective audio quality



# Room phone - microphones



4 super-cardioid microphones



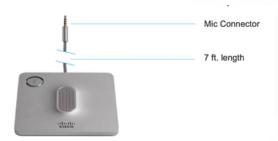
Suppresses echo physically and ensure 360-degree coverage

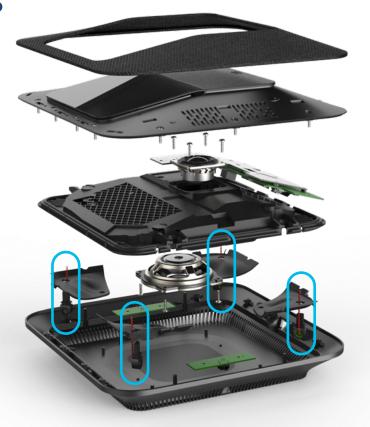


Frequency range from 100Hz to 12KHz.



2 omni-directional wired microphone







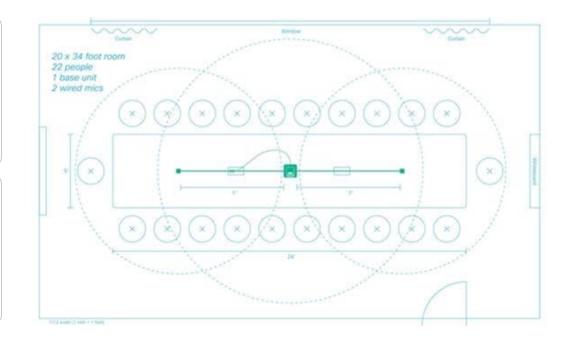
### Room phone- recommendations



The 4 super-cardioid microphones on the Room Phone are tuned to work best in rooms with upto to 10 people (20X20ft)



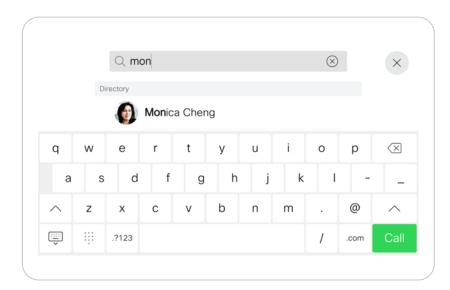
The 2 omni-directional wired microphones expands support to rooms with upto 22 people (20X34ft)

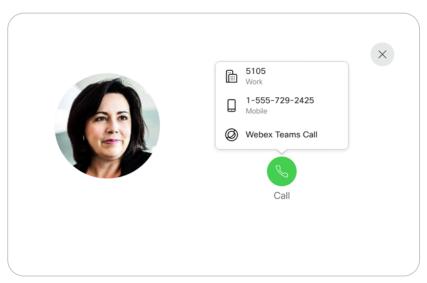




### Calling - search and dial

Easily find contacts and dial work or mobile numbers





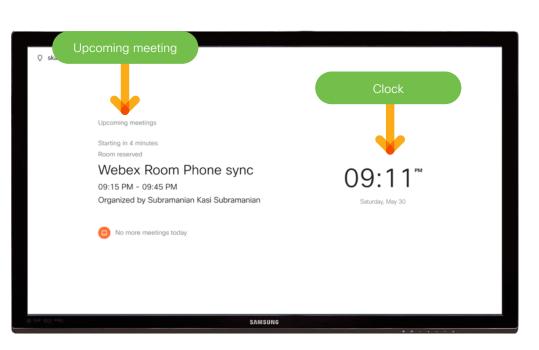
<sup>\*</sup> Requires WebEx activation and DirSync, onprem directories - roadmap



### Room calendar and one button to join

Quickly join scheduled meeting with one button





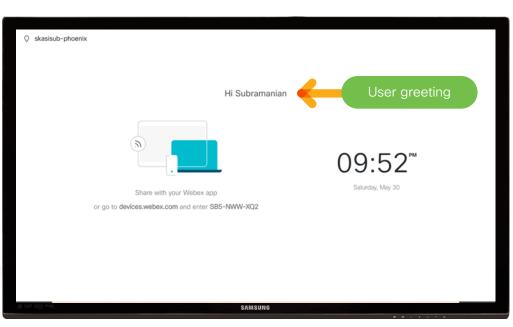
<sup>\*</sup> Requires WebEx activation and hybrid calendar



### Touchless meeting join

User greeted, automatic pairing using proximity and join through Webex apps on Laptop or Mobile device



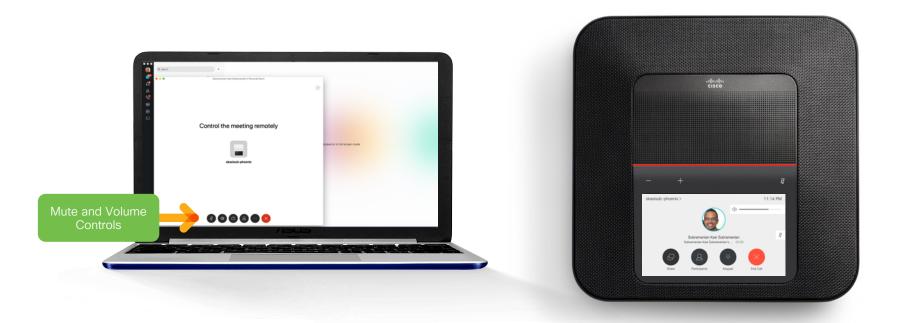


<sup>\*</sup> Requires WebEx activation and Webex meetings or Webex teams client



### Wireless in-meeting controls

Control mute and volume status using Webex apps



<sup>\*</sup> Requires WebEx activation, supported in Webex meetings or Webex teams client



### Active speaker and participant list

In-meeting context for collaborative meetings



<sup>\*</sup> Requires Webex activation, supported in Webex meetings



### Wireless sharing - Webex apps

### Share content wirelessly through Webex apps

- Share screen or specific application
- Option to Optimize for motion and video sacrifices some resolution in favor of a higher framerate.
- Share with audio, audio is played out through Room Phone speakers
- Device supports up to 1080p30fps (limited by senders/clients capabilities)



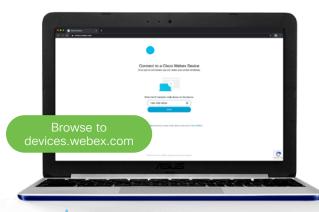


\* Requires Webex activation, supported in Webex meetings and calls. Sharing in onprem calls - roadmap

### Guest sharing

### Share content locally through web browser

- No Webex account required
- The pairing code shows on the external display
- Go to <u>devices.webex.com</u> and enter the code and the 4-digit PIN to confirm pairing





\* Device requires Webex activation



### Guest sharing

### Share content locally through web browser



Share screen, specific applications or browser tab



Option to optimize for motion and video sacrifices some resolution in favor of a higher framerate.



<sup>\*</sup> Device requires Webex activation



# Cable sharing

### Share using HDMI cable locally or preview and share in Webex meetings

- · Share content and audio locally
- Preview and share content with audio in a Webex call or meeting
- Capable of sharing up-to 1080p30
- Content audio is played out through Room Phone speakers

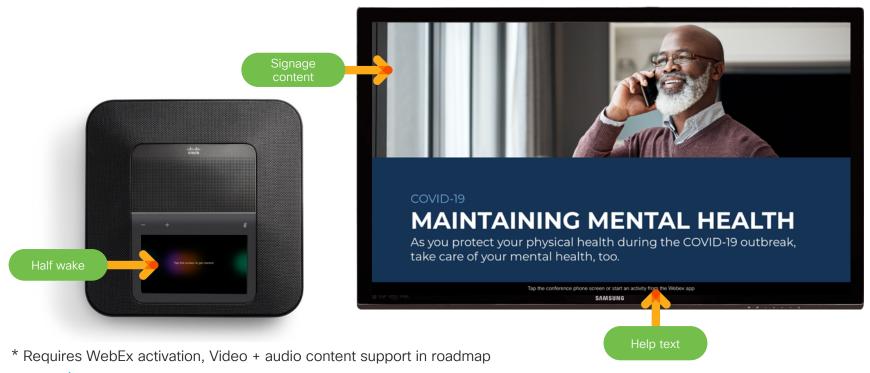




\* Requires WebEx activation, supported in Webex meetings and calls. Sharing in onprem calls - roadmap

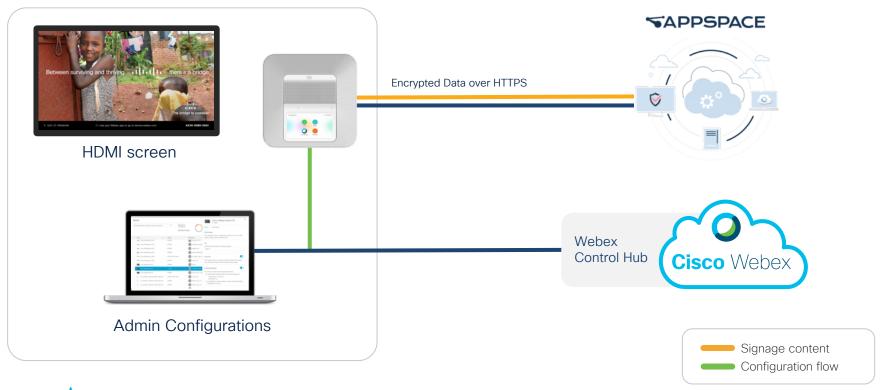
### Digital signage

Deliver content to screens (image carousel) through Appspace integration



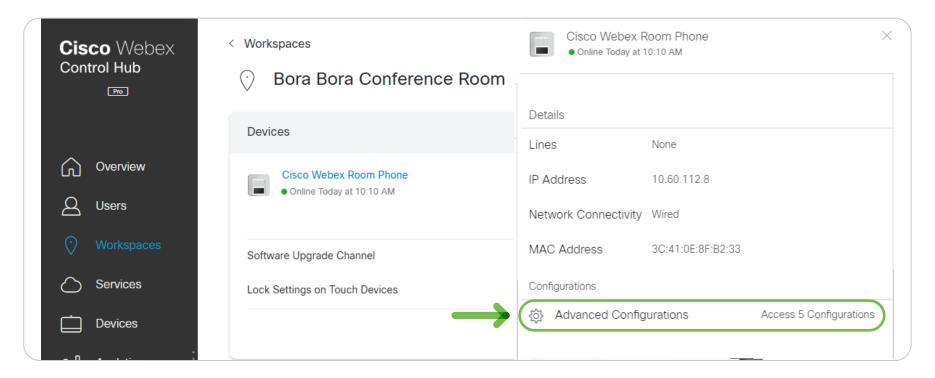
cisco life!

### Digital signage solution architecture



# Detailed configuration (Part 1)

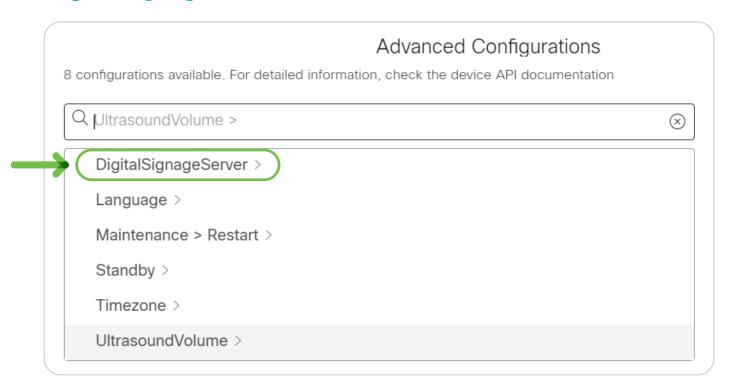
### In Webex Control Hub go to Advanced Configurations





# Detailed configuration (Part 2)

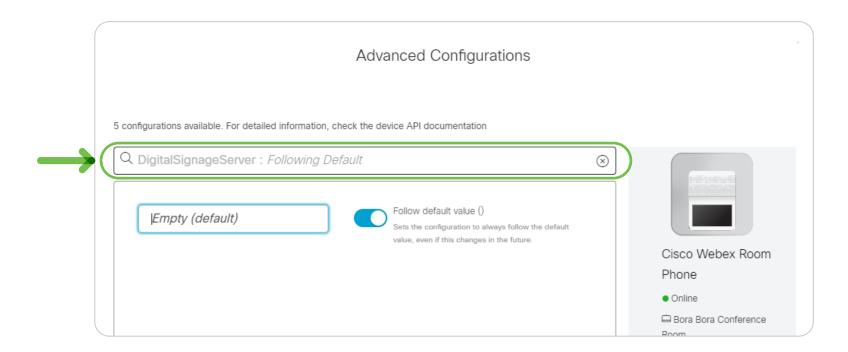
Select Digital Signage Server





# Detailed configuration (Part 3)

Populate the Digital Signage Server URL



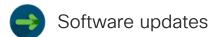


### Centralized management - Webex Control Hub

Provision new devices, monitor usage and troubleshoot call quality

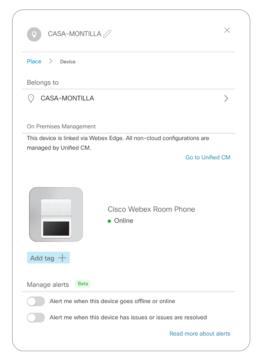
One pane of glass to view all your cloudregistered Cisco Room Systems





Push configuration

Get analytics









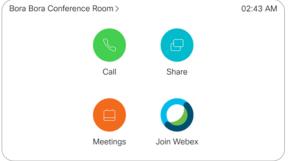
### Flexible deployment options

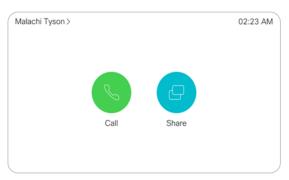
#### 1. Webex + UCM Calling

### 2. Webex + Webex calling

### 3. UCM Only









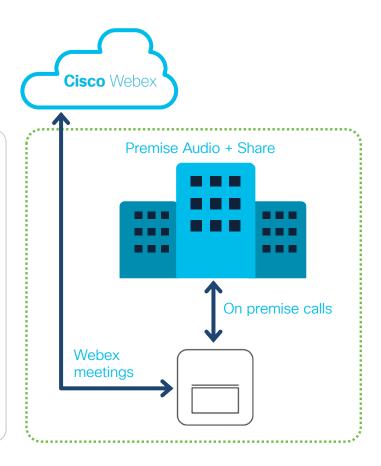






# Webex + UCM calling

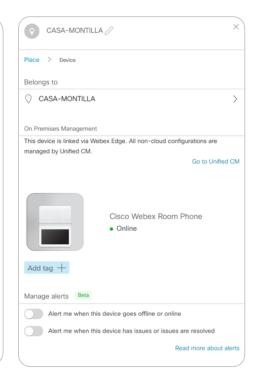
- Device Intelligently routes Webex meetings to the cloud and calls through UCM
- Brings the best of both worlds onprem calling and Webex meeting experience: Audio+Share (Participant list, active speaker, Lobby experience, in-meeting notifications)
- 3 Protect customers on prem investment
- No additional equipment required at customer premise





# Webex & UCM calling - onboarding and upgrade

- Use existing tools Control Hub and UCM Admin
- Enhanced device setup wizard to activate multiple services
- Manage device configuration through UCM and cloud configurations (calendaring, signage, proximity) through Webex control hub
- Automatic upgrades through control hub

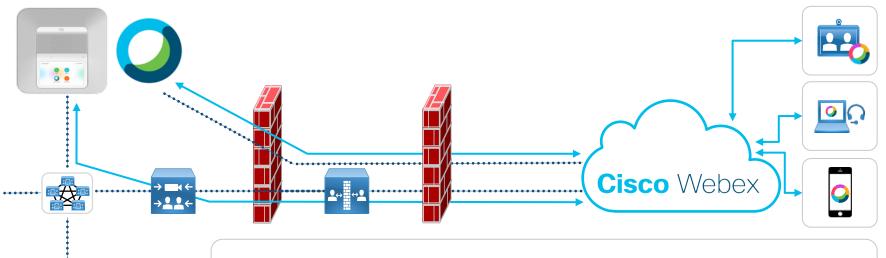






# Webex & UCM calling - call routing





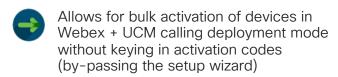
- Native Registration with Webex and UCM
- All non-SIP URI traffic (dialed digits) goes to CUCM and then can call on-premise devices or to PSTN
- Calls are routed to Webex if SIP URI dialed matching a WebEx domain (@webex.com, @wbx2.com, @ciscospark.com) or call originated from WebEx (OBTP, Paired Calls)
- All other SIP URI traffic routes to CUCM and optionally Expressway (dependent on SIP route patterns)
- Webex-only proximity sharing is permitted or use HDMI-in cable

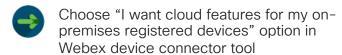


**PSTN** 

### Cisco Webex device connector

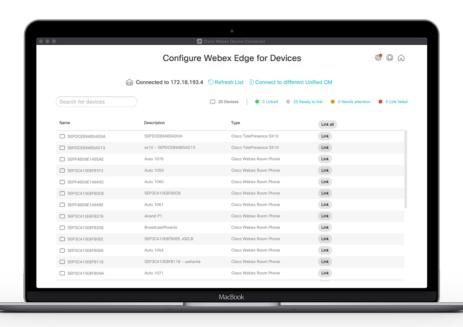
### Bulk activate UCM registered devices to Webex







Configure proxy in UCM under Enterprise Phone Configuration -> Product Specific Configuration Layout->"Proxy Settings for Webex" before linking if proxy is required to reach Webex

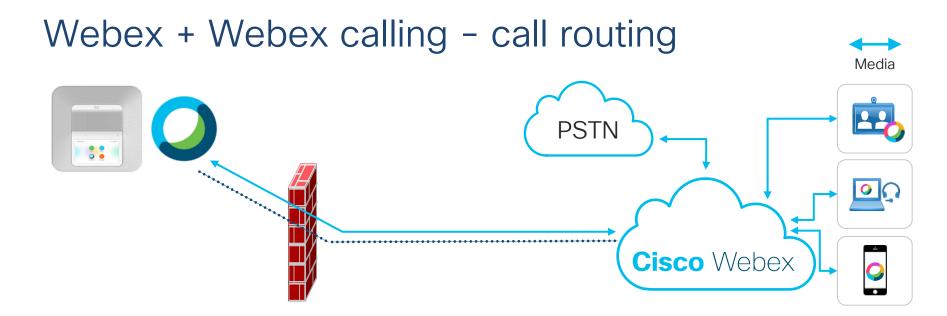


Proxy Settings for Webex

http://proxy.cisco.com:80









All non-SIP URI traffic (dialed digits) goes to Webex and then can call to PSTN



SIP URI traffic routes to Webex

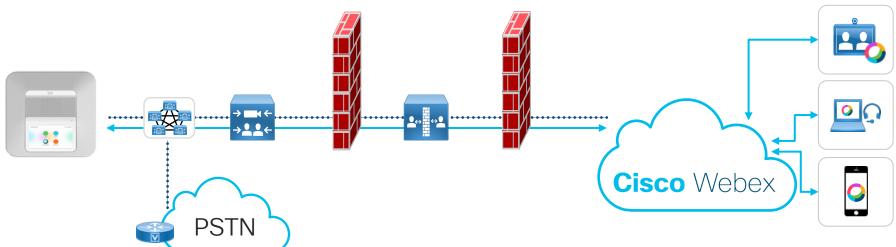


Webex calling support through control hub is available for PSTN access



# UCM only - call routing







All non-SIP URI traffic (dialed digits) goes to CUCM and then can call on-premise devices or to PSTN



SIP URI traffic routes to CUCM and optionally Expressway (dependent on SIP route patterns)



# Color options





### SKUs



#### CP-ROOM-C-K9=

Cisco Webex Room Phone (Carbon Black) Includes base unit, HDMI cables (3m and 8m), network cable



#### CP-ROOM-INJ=

Power injector for Cisco Webex Room Phone



#### CP-8832-MIC-WIRED=

Wired mics for Cisco Webex Room Phone



### Call to Action

- Evaluate firmware features in 12.7 and 12.8 on 78xx and 88xx
- Evaluate the Simple Phone Refresh functionality in CUCM 11.5 SU8 or CUCM 12.5 SU3
- Evaluate CUCM 12.5 SU1+ or CUCM 11.5 SU7+ with Cisco Headset 500 and 730 Series
- Evaluate the Webex Room Phone





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