





Unified Collaboration Architecture

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TECCOL-2982



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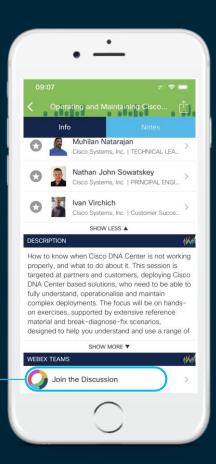
Cisco Webex Teams

Questions?

Use Cisco Webex Teams to chat with the speaker after the session

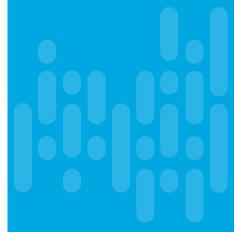
How

- 1 Find this session in the Cisco Events Mobile App
- 2 Click "Join the Discussion"
- 3 Install Webex Teams or go directly to the team space
- 4 Enter messages/questions in the team space



Agenda

- Introduction
- Preferred Architecture
- Clients Strategy
- Identity and Collaboration Tools
- Premise Architecture
- Hybrid Architecture
- Cloud Architecture
- Conclusions and Q&A



Collaboration























Phones







Headsets







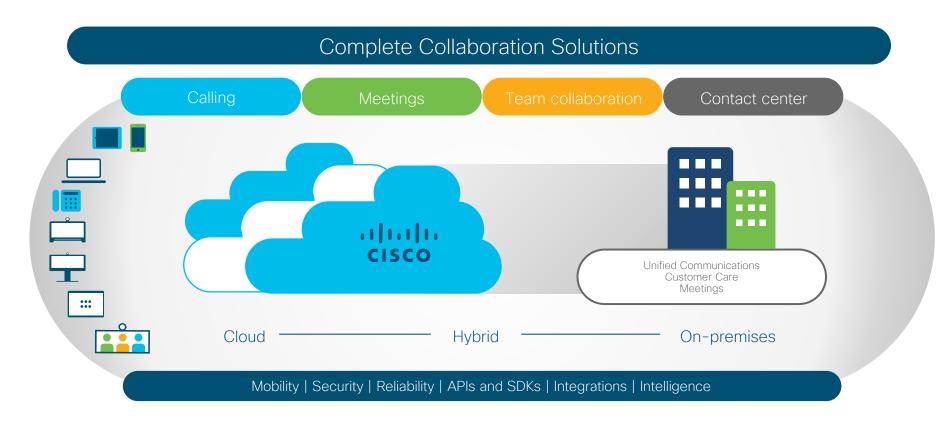




Preferred Architectures



Cisco Collaboration Architecture



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What are Cisco Preferred Architectures?

- 30-40 page Documents
- Prescriptive design guidance
- · Modular and scalable Design Base for Any Customer

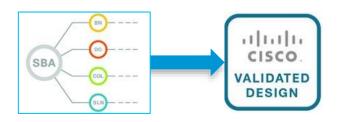
What products to use to enable users for Unified Communications for simple deployments?

Prescriptive Concise Tested best recommendations Documents practices



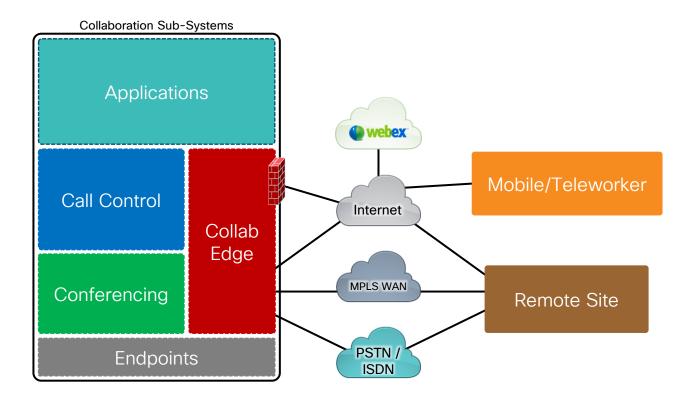
Existing Documentation

- Cisco Validated Design Guides (CVDs)
 - Two types of CVDs:
 - Technology Guides
 - Solution Guides
- Solution Reference Network Design (SRND)
 - 1,200 page comprehensive document covering network, endpoints, and all UC applications. Focus on design considerations and guidelines.
- Product Configuration and Deployment Guides
 - Step by step instructions for deploying specific products or solutions. Updated with each software release.





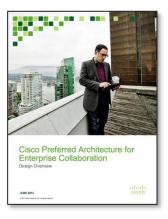
How are the Architectures Organized?

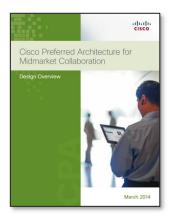


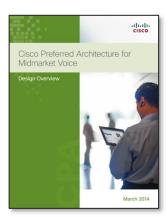


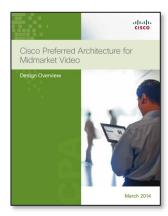
Cisco Preferred Architectures

- Cisco Preferred Architecture for Enterprise Collaboration
- Cisco Preferred Architecture for Mid-Market Collaboration
- Cisco Preferred Architecture for Mid-Market Voice
- Cisco Preferred Architecture for Video



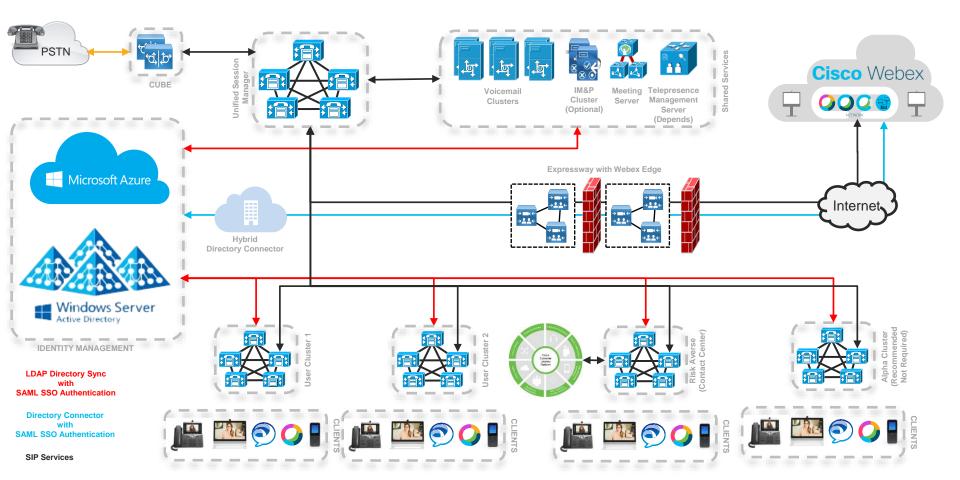








Preferred Architecture for Shared Services (Example)



Where can I get them?

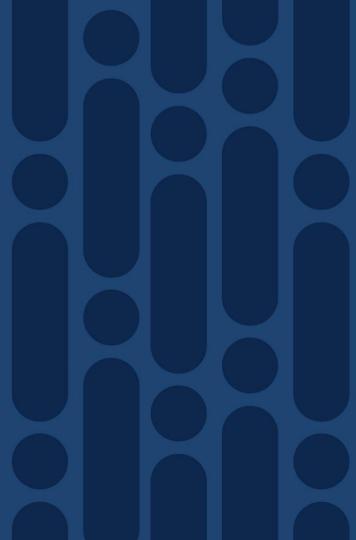
- Official URL:
- http://www.cisco.com/go/cvd/collaboration

- Shortcut:
- http://cs.co/pa4collab





Clients



The Clients Problem

Cisco has a clients problem, WHICH ONE





Clients Strategy

Choose Your Client

Premise (with a chance of cloudy)

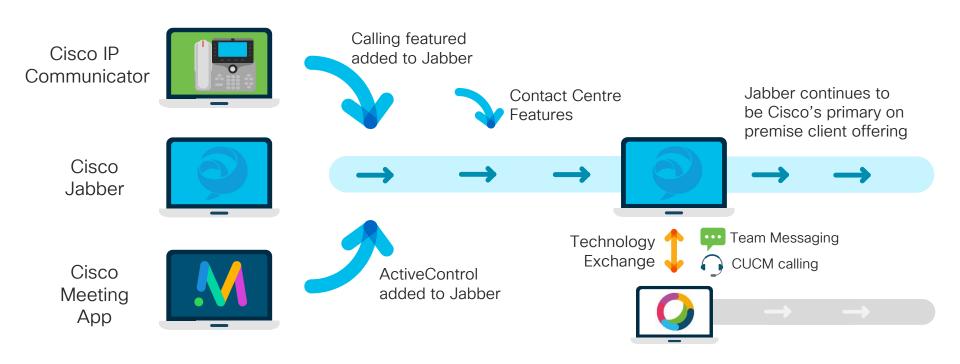


Hybrid and Cloud

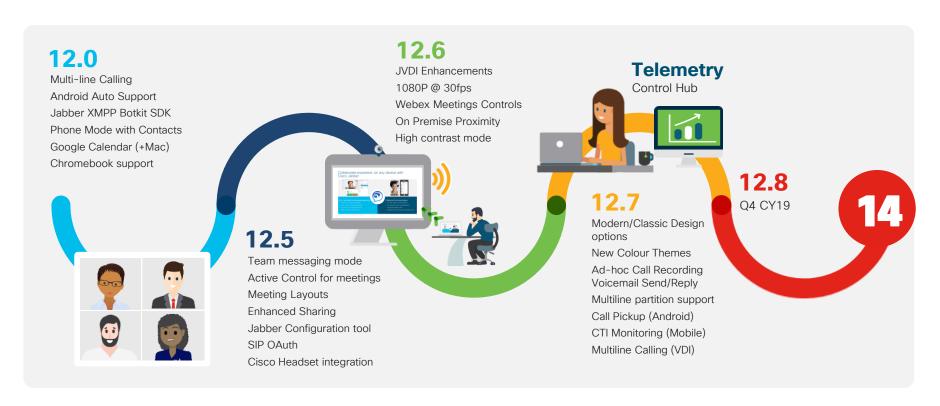




On Premise Client Rationalization



Jabber 12.x highlights so far....



12.7 feature summary



Application

- Modern/Classic Design options
- New Colour Themes inc Dark Mode



Calling & Meetings

- Ad-hoc Call Recording (Desktop)
- Voicemail Send/Reply (Desktop)
- Multiline partition support (Desktop)
- Call Pickup (Android)
- CTI Monitoring (Mobile)
- Multiline Calling (VDI)
- Audio Device selection policy (VDI)
- CMR Meeting Controls (Mobile)



Cognitive collaboration

People Insights



Messaging

Team Messaging

- Safari View SSO Support
- Enterprise Content Management
- Control Hub Config
- Search/Add BOT Contacts
- Authenticated Proxy Support
- Quote Message
- DND Duration

XMPP

Enhanced Security Labels

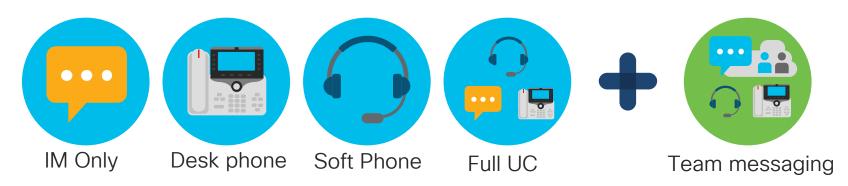


System/Devices

- Android 64bit Support
- OAuth/Logout Policy

Jabber operating modes

 Flexible operating modes means a organisation can configure the Jabber client for their needs...



 Jabber 12.5 + provides a new operating mode optimized for "team messaging"......

Why move to Jabber team messaging mode.....



Familiar User Experience

Limited retraining required to migrate to advanced messaging



UC Manager based Calling

Flexible calling options with strong feature set



Security and Encryption

Modern secure platform architecture



Persistent messaging

Persistent 1:1 and group messaging with search



Reduced Infrastructure

IM&P, P.Chat DB, MFT servers no longer required



What is Jabber team messaging mode

- New Jabber operating mode (single installer for all operating modes)
- A subset of Webex team messaging features are delivered by Jabber (XMPP not used)
 - Persistent 1:1 messaging
 - Space support
 - File support
- UC Manager / HCS rich call control features



Jabber team messaging mode Infrastructure

Webex team messaging service

IM&P/Messenger replaced by Webex team messaging, calling services provided by On Premise or HCS



Messaging Service

Presence Service

Contacts Service



Cisco Jabber

UC Manager/HCS

Unity Connection

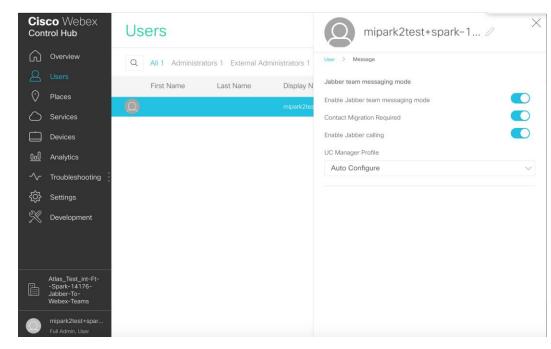
Corporate Directory

Mobile Remote Access



Team messaging: Control Hub

- Jabber support in Webex Control Hub
 - Service discovery
 - · Contact Migration
 - · UC manager profile



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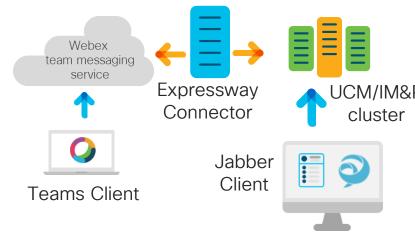


Is this the same as Jabber/team messaging Interop....?



....Simple answer is NO,

team messaging mode is a Jabber native registration to Webex team messaging platform



 Jabber/Webex team messaging interop is a XMPP/team messaging gateway service provided on Expressway



In The Beginning!

There Was ONLY One Client





Webex Teams - Modular Client

Meet customer needs in a modular approach



Flexible Designs

- ✓ Single app Single or multiple workloads
- ✓ Address compliance/security issues with Microsoft or Slack
- **✓** Support 3rd party integration
- ✓ Single Admin experience, align to any workstyle via Control Hub



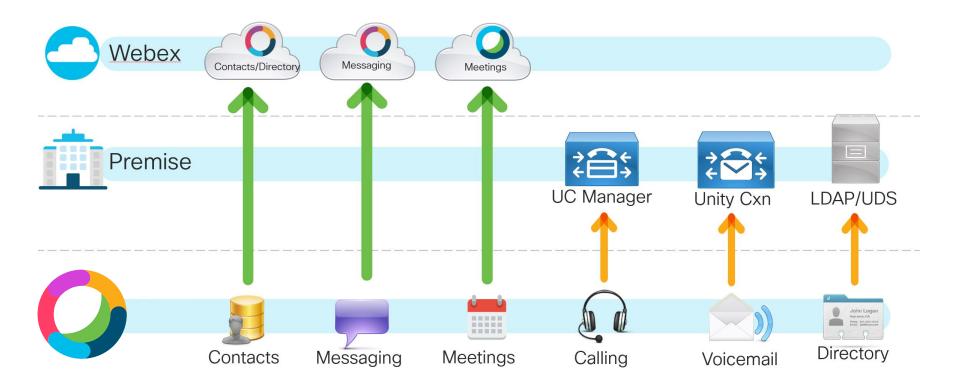
Cisco Webex Calling



Unified CM (On Premise, HCS, UCM Cloud)



Webex Teams Client Architecture - Hybrid



Identity
One Source To
Rule Them All



Common Identity Overview





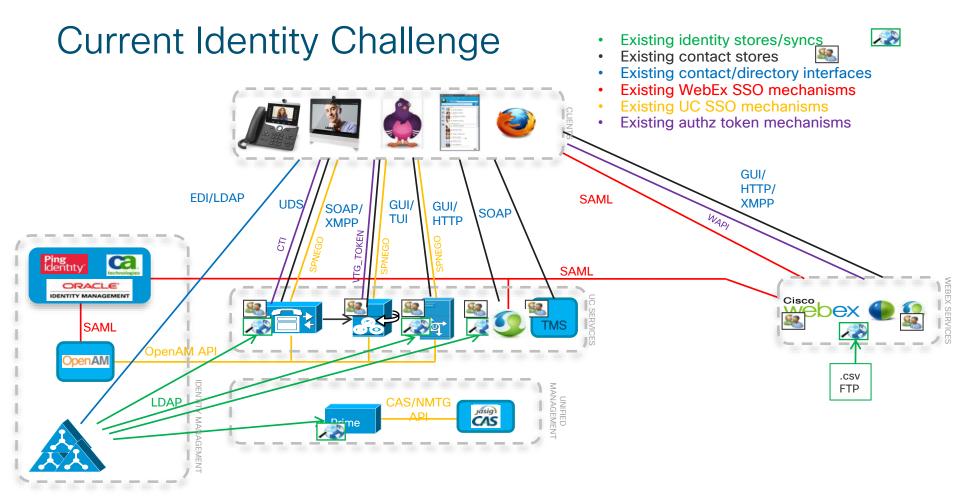




Three Components of an Identity System

- Account
 - An individual's identifier: should be globally unique
 - Typically, this does not change
 - For example, Employee ID or User ID
- Authentication
 - Mechanism used to identify the user is who they say they are
 - Can be based on any attribute supported by the directory vendor or the Identity Management System (IdMS)
 - Examples:
 - Active Directory supports authentication using SAM Account, UPN, or user logon name
 - IdMS can support SAM Account, Employee ID, Email, etc.
- Authorization
 - A users ability, or permission, to access resources or services





What Does All This Mean?

Identity Matters!

- Cisco Collaboration Solutions are not owners of identity, they are consumers
 - No password change or enforcement capabilities
- When considering multiple platforms (both premise and cloud) it is important that the primary email address matches across all platforms, including discovery addresses on Cisco cloud platforms!
- Single Sign-On and Edge solutions require unique Discovery ID's (BOB@CISCO.COM) across the entire portfolio that match in all systems if you want to use all of the auto provisioning and placement tools



Identity Considerations

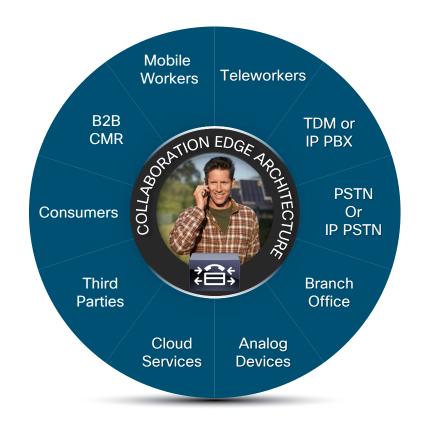
Why the concern

- Communications Manager is capable of ingesting multiple domains, but can only point to two authentication source
 - All domains should be consolidated into a single digest system for authentication
- Communications Manager has a character limit on the domain field (255 characters)
 - May require multiple cluster depending on number and length of domains
- For multiple domain situations, Communications Manager 10.5.2 or later is required for Flexible JID
- Cloud integrations with Webex require the user primary email address match all discovery alias



Identity Breakdown

- -All users have a globally unique identifier (GUID)
- All other attributes are defined as variables under the user account on the IDP
 - Account or Logon
 - Telephone Number
 - 10 or 11 digit
 - -E.164
 - -+ Number ID
 - Email Address
 - -IM Address
 - SIP Address



Identity Workflow





Standard Attributes
Contained in Identity (NY.GOV)

- ID Emp ID or GUID
 - Account
 - Telephone #
 - SIP Address
 - IM Address
 - Email Address



Agency Microsoft Domains and Services ITS, TAX, ETC

Cloud Services



Collaboration Tools







One Name to Rule Them All

Know The Rules

- If you want SSO to work across the board
- Globally unique user-id should be used (there can be only one bob!)
- Use a SAMLv2 IdP for all your local and cloud systems (Oasis Standard)
- Ensure all data is in your IdP, and ensure that it is correct
 - Unique discovery address
 - Telephone Numbers (E.164 or + Number format recommended)
 - Proper Group Access and Assignment
- IdP must be accessible locally as well as from the internet



Single Sign On, SSO

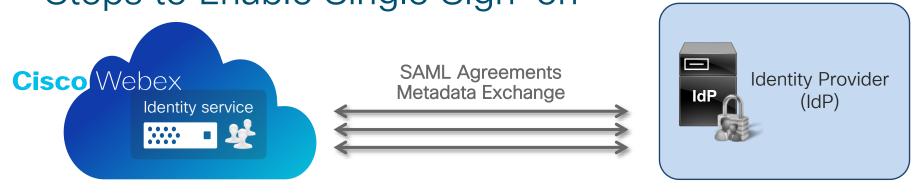


SSO - Cisco Premise and Cloud

- Cloud based SSO integration is normally configured in Control Hub
- Supported SAML 2.0 requirements, but we do not need multiple ACS capabilities that we do need for premise based Cisco architectures.
- Due to the multiple ACS requirement of premise based Cisco applications (CUCM)
 we do not support many cloud based SSO platforms due to the fact they do not
 support true multiple ACS redirects (Google and Microsoft AzureAD)
- Other Cloud platforms are supported with premise based OKTA, Shibboleth, Duo
- Control Hub and non-Cl Webex Site admin sites support cloud to cloud SSO configurations.



SAML SSO Configuration for Webex in Control Hub Steps to Enable Single Sign-on



Collaboration Admin Identity Admin

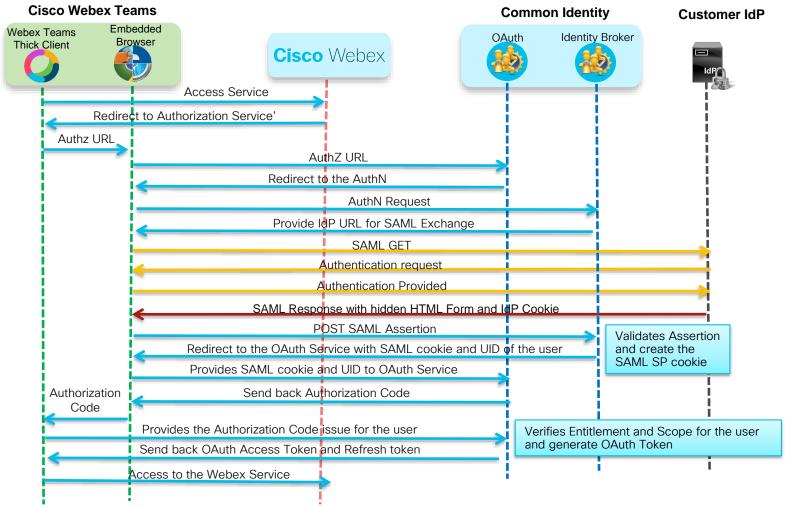
- Download SP Metadata
- Integrate With 3rd Party Identity Provider
 - Choose Signed or not Signed Metadata
 - Upload IdP Metadata
 - Test SSO Connection

cisco Live!



Download IdP Metadata





```
<saml2p:Response
Destination="https://idbroker.webex.com/idb/Consumer/metaAlias/ea7c1420-
711d-4916-95f8-22de53230d1e/sp"
          ID=" 157561492b8068bb78f4cb242ad4f006"
InResponseTo="s2e747a3b284812b71ccd8ac1dce98d00cbfa7555b"
          IssueInstant="2017-01-30T17:13:22.572Z"
          Version="2.0"
          xmlns:saml2p="urn:oasis:names:tc:SAML:2.0:protocol"
  <saml2:lssuer Format="urn:oasis:names:tc:SAML:2.0:nameid-format:entity"
          xmlns:saml2="urn:oasis:names:tc:SAML:2.0:assertion"
           >https://shib9a.cisco.net/idp/shibboleth</saml2:lssuer>
  <saml2p:Status>
     <saml2p:StatusCode Value="urn:oasis:names:tc:SAML:2.0:status:Success"/>
  </saml2p:Status>
   <saml2:Assertion ID=" 574a68c9ba24935315c606a48902e50f"</p>
            IssueInstant="2017-01-30T17:13:22.572Z"
            Version="2.0"
            xmlns:saml2="urn:oasis:names:tc:SAML:2.0:assertion"
            xmlns:xs="http://www.w3.org/2001/XMLSchema"
    <saml2:lssuer Format="urn:oasis:names:tc:SAML:2.0:nameid-</pre>
```

format:entity">https://shib9a.cisco.net/idp/shibboleth</saml2:lssuer>

```
<ds:Signature xmlns:ds="http://www.w3.org/2000/09/xmldsig#">
       <ds:SignedInfo>
         <ds:CanonicalizationMethod Algorithm="http://www.w3.org/2001/10/xml-exc-</p>
c14n#" />
         <ds:SignatureMethod Algorithm="http://www.w3.org/2000/09/xmldsig#rsa-sha1" />
         <ds:Reference URI="# 574a68c9ba24935315c606a48902e50f">
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signature" />
              <ds:Transform Algorithm="http://www.w3.org/2001/10/xml-exc-c14n#">
                <ec:InclusiveNamespaces PrefixList="xs"
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            </ds:Transforms>
            <ds:DigestMethod Algorithm="http://www.w3.org/2000/09/xmldsig#sha1" />
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         </ds:Reference>
       </ds:SignedInfo>
       <ds:SignatureValue>L0n0SdlaXFyL4Eg6......</ds:SignatureValue>
       <ds:KeyInfo>
         <ds:X509Data>
               <ds:X509Certificate>MIIDKzCCAhOgAwlBAgIUNXw.......<ds:X509Certificate>
         </ds:X509Data>
       </ds:KevInfo>
     </ds:Signature>
```



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```
<saml2p:Respense
Destination="https://idbroker.webex.com/idb/Consumer/metaAlias/ea7c1420-
711d-4916-95691293e5930theylebex
          ID="_157561492b8068bb78f4cb242ad4f006"
InResponseTo="s2e747a3b284812b71ccd8ac1dce98d00cbfa7555b"
          IssueInstant="2017-01-30T17:13:22.572Z"
          Version="2.0"
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  <saml2:lssuer Format="urn:oasis:names:tc:SAML:2.0:nameid-format:entity"</p>
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             Version="2.0"
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             xmlns:xs="http://www.w3.org/2001/XMLSchema"
     <saml2:lssuer Format="urn:oasis:names:tc:SAML:2.0:nameid-</pre>
format:entity">https://shib9a.cisco.net/idp/shibboleth</saml2:lssuer>
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c14n#" />
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              </ds:Transform>
            </ds:Transforms>
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       </ds:SignedInfo>
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         <ds:X509Data>
               <ds:X509Certificate>MIIDKzCCAhOgAwlBAgIUNXw.......<ds:X509Certificate>
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       </ds:KevInfo>
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```



```
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InResponseTo="s2e747a3b284812b71ccd8ac1dce98d00cbfa7555b"
          IssueInstant="2017-01-30T17:13:22.572Z"
          Version="2.0"
          xmlns:saml2p="urn:oasis:names:tc:SAML:2.0:protocol"
  <saml2:lssuer Format="urn:oasis:names:tc:SAML:2.0:nameid-format:entity"
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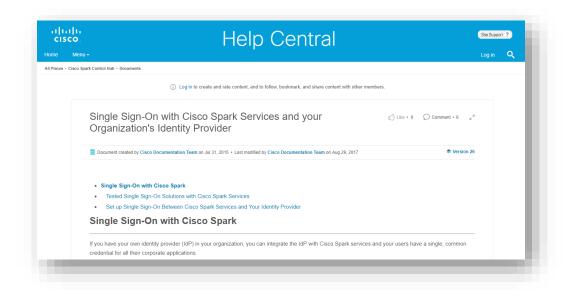
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          xmlns:saml2="urn:oasis:names:tc:SAML:2.0:assertion"
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           <ds:Transforms>
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signature" />
              <ds:Transform Algorithm="http://www.w3.org/2001/10/xml-exc-c14n#">
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                                     Webex to validate
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         </ds:X509Data>
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```



Cloud Collaboration SSO integration guides

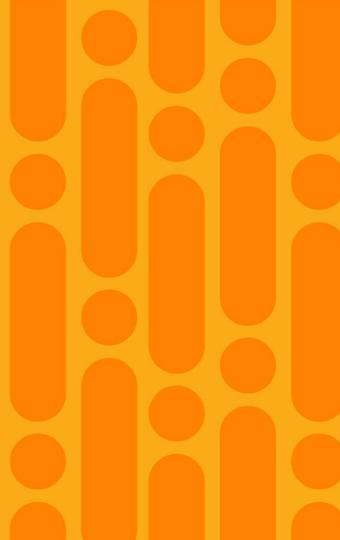


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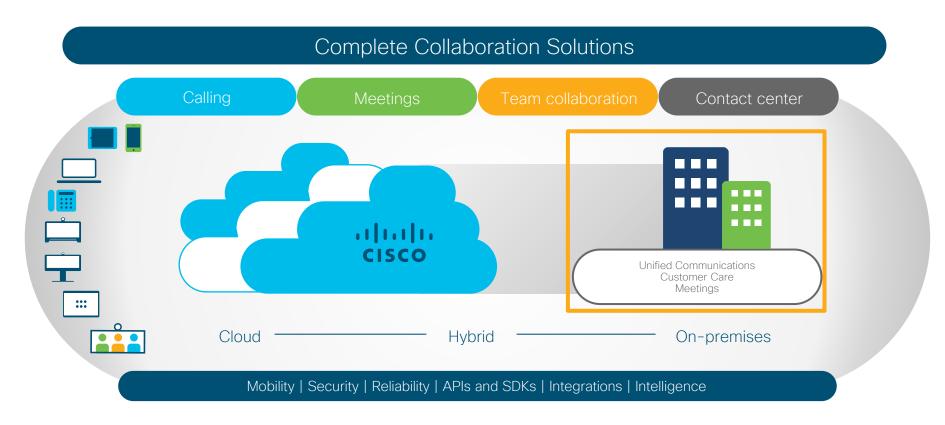


Premise Architecture

with a chance of Cloudy!



Cisco Collaboration Architecture





Agenda

- Call Control and Core Services
- Conferencing
- Applications
- Remote Access



Call Control and Core Services



What is a Cluster

A CUCM cluster is the brains of the premise-based call control platform, made up of:

- 1 Publisher (only one publisher per cluster)
 - Standard Cluster Up to 8 subscriber nodes supporting up to 40K (1:1) or 80K (No Redundancy) SIP/SCCP encrypted/non encrypted devices
 - Mega clusters of up to 16 subscriber nodes supporting up to 80K (1:1) or 16K (No Redundancy) SIP/SCCP encrypted/non encrypted devices (requires design review by support)
- Additional nodes for TFTP and media services can be added and recommended in clusters over 1250 devices



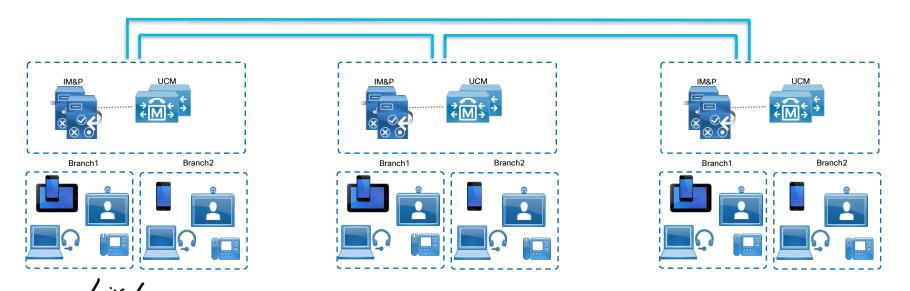
Clustering Behavior and Caveats

- Latency between campus deployed nodes should not exceed 15ms
- Latency between wan deployed cluster nodes should not exceed 80ms
- Latency between end user devices and centralized services should not exceed 150ms
 - Any device that exceeds more than 200ms can experience issues such as delay-to-dialtone or increased media cut-through delays



Call Control Architecture and Deployment Models: Simplification

- Recommendation: Centralized Call Processing Model (Single Call Processing Cluster)
- Full-Mesh Distributed Call Processing Deployment Model (Multiple Call Processing clusters)
 may be required. This model is based on multiple iterations of the Centralized Call
 Processing Deployment Model



Call Control Benefits

- Single call control for voice, video, IM&P
- Collaboration-friendly dial plan which makes easy to add video to voice, IM&P to voice and video
 - Simplified deployment model, design, dial-plan, video, IM&P integration, sizing, etc.
- Modular architectural approach which enables better scalability
- Add additional services avoiding re-configuration costs



Single Cluster or Multi Cluster

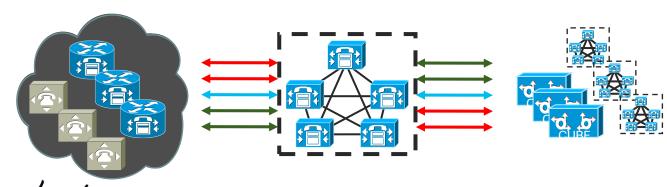
When and why to consider

- Multiple Domains
 - Because many of the systems are not multi-domain capable for all functions and not all customers can deploy an IdP or consolidate domains
- Risk Averse Lines of Business
 - If you have critical business services that cause you to not upgrade on normal schedules because of fear of change or business impact
- International or High Latency Links
 - Any time latency between sites exceeds recommended deployment guidelines for real time communications or database replication
- Alpha clusters for user acceptance and testing HIGHLY RECOMENDED



What is a Session Management Edition Cluster

- A CUCM cluster and a SME cluster use exactly the same software
- A CUCM cluster is typically used to register phones
- A SME cluster is typically used as a platform for Trunk and Dial Plan aggregation
- Both CUCM and SME support Voice, Video, and Encrypted calls



Reasons for Deploying a SME Cluster

Functions that a SME Cluster Offer

· Centralized Dial Plan

- Globalize and Normalize the called and calling numbers used by al leaf systems
- Mange overlapping number ranges in Leaf systems
- Configure "find me" call routing using Route List and Route Groups
- Re-route calls via PSTN when the device cannot be reached via IP path
- · PSTN trunk consolidation and dynamic re-routing

Mobility Features

- Offer Single Number Reach functionality to devices on 3rd Party UC systems
- Offer Extend and Connect functionality to devices on 3rd Party UC systems
- Normalization Scripting
 - Allow you to modify inbound and outbound SIP message and SDP body content – simplifying interoperability with 3rd party UC systems



Features to Utilize in a SME Cluster Design

- Extension Mobility Cross Cluster (EMCC)
 - Public Space Phones
 - Contact Center Agents
 - Hoteling
- TFTP (proxy TFTP Server)
 - Allows all endpoints to point to a single TFTP server and be redirected to the correct "home server" for config and registration
- PSTN Access
 - PSTN Access for all cross cluster members

- Inter-Cluster Lookup Service (ILS)
 - Allows CUCM to learn about remote clusters without the need for an administrator to manually configure connections between each cluster
 - ILS URI Replication feature enables ILS to exchange directory URI catalogs with the other clusters in an ILS network. URI Replication provides support for intercluster URI dialing
- · RSVP Agent
 - Allows local vs. home cluster media resources for EMCC devices
- Location Bandwidth Manager (LBM)
 - Allows for full-mesh replication of their cluster topology



Communications Manager Architecture

The center of the collaboration experience

- All current and capable shipping endpoints register to CUCM
 - · Soft-clients, Personal Endpoints, and Room Systems register here!
- Single SIP domain (cluster wide setting)
- CUCM is first and always is a directory number-based system
- Always use FQDN
 - · Put Hostnames, DNS servers, and Domains on systems when upgrading or installing
 - Install certificates on all internal and public facing systems (use a trusted CA, not self-signed or internal CA)
 - Turn ON Encryption on all UC infrastructure!!



Dial Plan Getting Started

- Best and most important tools for dial plan design:
 - Pencil
 - Paper
 - Whiteboard
- Dial plans are not a new concept
- IP did not change the fundamentals of dial plan design
- Dial Plan recommendations are not a monolith
 - Take what you need
- Keep it simple!



What Is a Dial Plan?

- Mapping from dialed destinations to connected endpoints
- Concepts that are part of dial plans
 - user input
 - mapping of user input to routable format (transformations)
 - routing / routing restrictions (class of service)
 - call presentation
 - numbering plans





User Input / Dialing Habits

- Dialing habits are the common ways a user dials a number
- Different formats for types of destinations
 - colleague next door 4XXX four digit dialing
 - local, national, international 9-7D, 91-10D, 9011-E.164, +E.164
 - Inter-office (abbreviated on-net, forced on-net) 8-7D
 - Voicemail 4000 special intra-site dialing
 - Emergency
 - other services
- Especially external dialing habits are country-specific
 - 9 or 0 for outside line
 - Format of national numbering plan (fixed/variable length etc.)



Enterprise Specific Dialing Habits

- Dialing habits for local, national, international calls are typically given/agreed upon based on a given domain/country
- In addition need to agree on how to dial:
 - Private numbers (on-net)
 - Intra-Site
 - Services (voicemail, meet-me, call park, pick-up ...); non-DIDs
- "+" dialing also needs to be supported!
 - application support
 - number portability



Dial Plan vs. Numbering Plan

- Dial Plan: from dialed digits (dialing habits) to destinations
- Numbering Plan: scheme to number entities (phones and services)
 - unique number per entity → e.g. (+)E.164, private numbering
 allows for single numbering domain
 - overlapping numbering → e.g. unique per site
 requires partitioned numbering domains
 (and site-specific partitions/CSS, even w/ LRG)
- Recommendation: unique addresses
- Benefits of unique addresses
 - Maintain correct caller ID (think overlaps in forwarded inter-site calls)
 - Simplified VM integration (unique subscriber IDs)



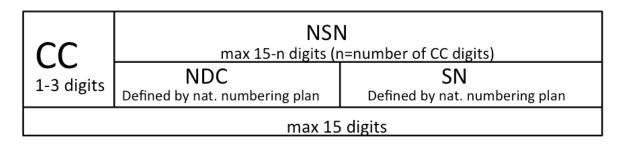
Dial Plan vs. Numbering Plan (NANP Example)

- · Dial plan might support various dialing habits
 - local call: 9 number
 - national call: 91 number
 - international call: 9011 number
 - abbreviated on-net: 8<7-digits>
 - +E.164: +E.164 string
- Enterprise Numbering Plan might follow one of the above dialing habits (e.g. abbreviated on-net)

... but does not necessarily have to!



E.164 geographic numbers



ITU Recommendation E.164 describes the "Numbering Plan of the International telephone service"

- CC: Country Code
- NSN: National significant number
- NDC: National destination code
- SN: Subscriber number
- NDC+SN = NSN: National significant number

National numbering plan left to national authorities

- documented at http://www.itu.int/oth/T0202.aspx?lang=e <a href="nage-embedding-nage-embed-embedding-nage-embedding-nage-embedding-nage-embedding-nage-emb
- US: fixed length, NSN 10 digits
- DE: variable length, NSN 4-13 digits



+E.164 Notation and "Numbers"

- ITU Recommendation E.123 describes the "Notation for national and international telephone numbers, e-mail addresses and Web addresses"
 - "+" signifies the international prefix
 - Example: +14085551234
- Numbers in global directories should be in +E.164 format
 - global form including country code
 - · leading "+"
 - no trunk access codes included: +44 (0) 208 1234 1243 is NOT a valid +E.164 number!
 - universal use
- Benefits of +E.164 "Numbers" in dial plans
 - unique by definition
 - no overlap with any other dialing habit ("+")



+ Sign Support

What It Is: Concept

- +E.164 support includes the use of + to wildcard international access codes AND to avoid overlap between globalized numbers and other ranges (e.g.: calls to India (+91XXXXXXXXXX) and NANP toll calls (912125551234)
- Supporting the + sign allows UCM-based systems to route calls based on an universal non-site (country) specific format
- + can be used in all dialable patterns
 - DN
 - Route Pattern
 - Translation pattern
- Most phones support +-dialing: 7925/21 from day one, newer phones starting with phone firmware 9.1.1



+ Sign Support

Caveats

- 1st generation phones (e.g., 7940/60) do not support + dialing from phone directories
- Unity Connection 9.0 supports +E.164 subscribers
 - +E.164 alternate extensions, MWIs supported with 8.x
- Emergency responder 10.0 supports +E.164
- Contact Center Express
 - +E.164 CTI route points and CTI ports (UCCX 8.5(1)SU3)
 - +E.164 agent extensions (UCCX 10.0 Finesse only)



Directory and Onboarding



Directory, Where It All Starts!

Identity Matters!

- Cisco Collaboration Solutions are not owners of identity, they are consumers
- Supported LDAP Integrations
 - Microsoft Active Directory 2003/2008/2016
 - Microsoft Lightweight Directory Services (LDAP)
 - Sun ONE Directory
 - Open LDAP
 - Oracle Directory Server Enterprise Edition



Self-Provisioning

Administrative Improvements

User administration has been greatly simplified!

Makes use of Profiles and Templates

User Profile, Service Profile, Device/Line Templates

Templates can now be applied at the LDAP integration or at time of user provisioning.

Two provisioning mechanisms added

IVR-based Self Provisioning (No Administration Required)

Quick User/Phone Add





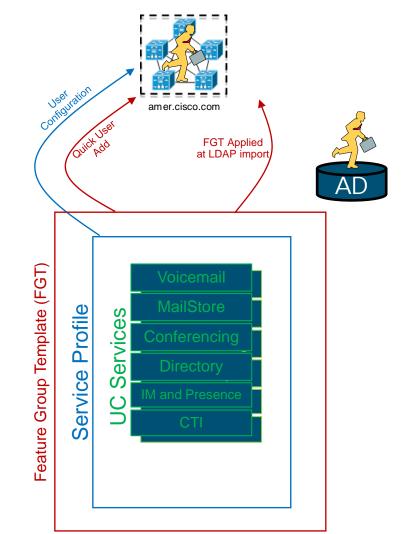


Self-Provisioning Foundation

Service Profile Concept

- The Service Profile created to simplify Jabber registration and configuration
- A Service Profile is made up of UC Services. Each UC Service represents a Jabber workload:
 - Voicemail, HTTP connection for Jabber for visual voicemail
 - Mail Store, IMAP connection for Jabber to Unity Connection for voicemail retrieval
 - Conferencing, WebEx site information or Jabber to WebEx or CWMS
 - Directory, TCP connection for Jabber to AD for LDAP Contacts (as opposed to UDS)
 - IM and Presence, XMPP connection information for Jabber to CUCM IM/P
 - CTI, TCP connection for Jabber to CTI application server (CUCM phone control)
- For Redundancy, there can be more than one UC Service for a given Jabber workload (this depends on the workload application)
- The UC Services are assigned to one or more Service Profiles
 - Primary, and sometimes Secondary, and Tertiary UC Services are assigned to a Service Profile
- The Service Profile is assigned to the user by way of
 - The User's configuration page
 - Through a Feature Group Template (FGT) either...

directly (via Quick User Add) or indirectly (via LDAP integration)



Self-Provisioning Foundation

Provisioning Configuration

Provisioning Configuration "Deconstructed":

Assign **Access Control Groups** to the LDAP Integration configuration

Can **Create and assign directory numbers** from LDAP (with mask) or from a pool of numbers. The DNs will not be assigned to a device until the device is provisioned

Assigned **Feature Group Templates (FGT)** to LDAP integration. Within FGT, we...

Assigned Home Cluster and enable IM and Presence. **Do not assign a single user to multiple clusters**

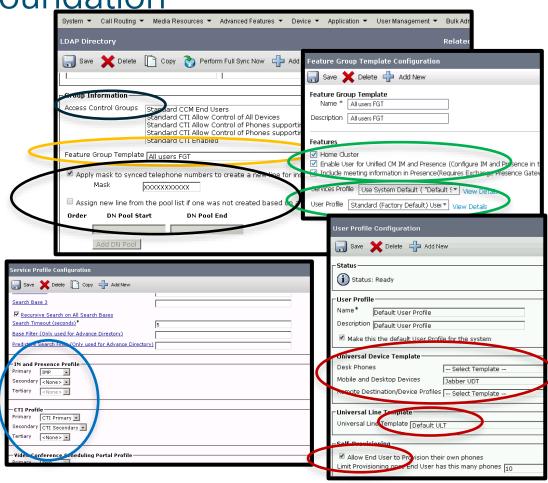
User Profiles and Service Profiles

User Profile (Universal Device and Line Templates not shown)

Turn On Self Provisioning

Add Services to the Service Profile

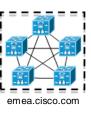




Self-Provisioning

LDAP and Provisioning with Multiple Clusters

- User's must be provisioned on only 1 cluster, so provisioning users at LDAP import needs to grab only users that belong to it.
- This can be done in the following ways:
 - Direct integration's User Search Base at different OU's. This might require a somewhat geographical OU design, which is unlikely.
 - Standard I DAP filters
- With Standard LDAP Filters, you must have attributes to filter on
 - LDAP filters can be nested and complex in order to zero in on the specific user LDAP attributes, but your AD must have the necessary user information in order.
- You can create a DN at import using Feature Group Templates. Phone number population in AD is important
 - If in full +E.164 from AD, then you just need to accept that as the DN
 - URI assigned from mail or msRTCSIP-primaryuseraddress AD attribute
 - Enable Routing (ILS) and globalized addressing. If imported phone number field not populated with full +E.164, might be much harder to create the Enterprise Alternate Number
- Assign Home Cluster on import
 - Home cluster required for IM/P server assignment and URI blended Identity creation
 - Jabber with IM/P up and running, zero touch
- In 10.x, we have increased the number of LDAP integrations (to single AD Forest) in a cluster to 30
 - Might be a trade off of # integrations vs. filter complexity



(&(objectclass=user)(I(co=United Kingdom)(co=France)))

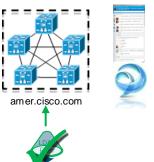




Directory Number +14085551212 Blended URI Enterprise E.164 # w / ILS Enterprise Alt. # w/ ILS **Home Cluster**

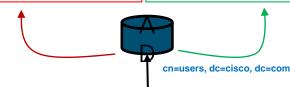
Create the following using FGT

idoe@cisco.com +14085551212 85551212 **Assigned**





(&(objectclass=user)(l(co=United States)(co=Mexico)))







Attribute sAMAccountName

telephoneNumber otherTelephone

Value idoe United States idoe@cisco.com +14085551212

85551212

Self-Provisioning

Creating the Device

Two Ways to Provision Device using CUCM 10 Provisioning

1. Self Service

Used in conjunction with Auto Registration

Phone must be auto registered before provisioned

TIP → add a speed dial to the Universal Line Template assigned to auto auto-registered devices

Self Service with CUCM IVR

Self Service IVR must be created first on Publisher

User follows prompts enters Self Service ID

Self Service User ID = form of user phone number

2. Quick User/Phone Add

Manually add device to local or AD user

Can apply User Info, FGT and auto-created DN automatically from LDAP import

Provisioning might be limited to LDAP capabilities

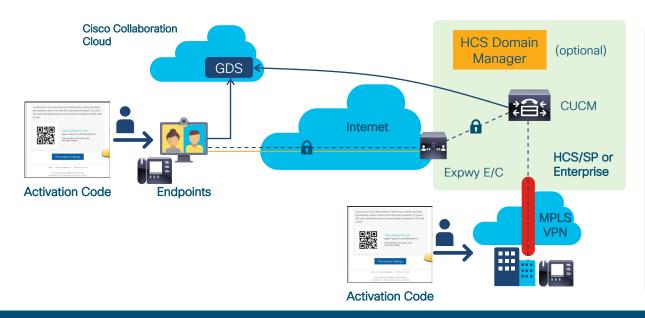
UCM 10.x now supports 30 LDAP integrations

LDAP attributes and filtering extremely important in multi-cluster environment





Activation code-based device onboarding



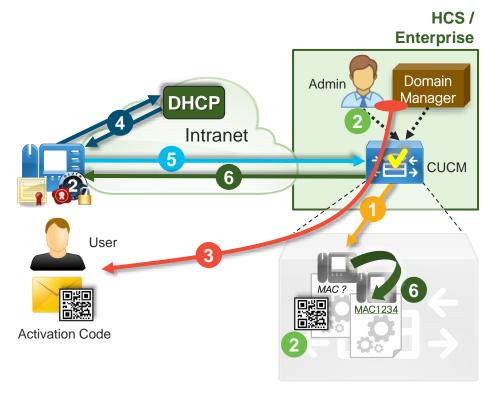
- Simplifies user experience for device onboarding
- Faster onboarding process reduces deployment costs
- Secures Internet access, allowing broader adoption for MRA

78xx and 88xx IP Phones - 7811, 7821, 7832, 7841, 7861, 8811, 8841, 8845, 8851, 8851NR, 8861, 8865, 8865NR

^{*}Supports on-premises deployments with 12.5 Mobile and remote access using GDS requires 12.5 SU1



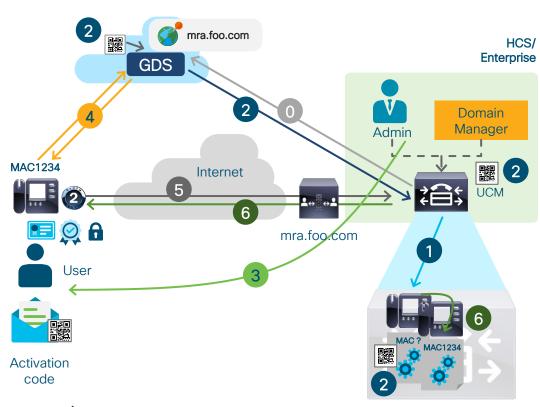
Activation code-based onboarding - Onpremises Flow



- 1 Admin creates full device config without specifying MAC address
- 2 Admin requests activation code for this device (CUCM-generated)
- 3 Activation code sent to a user/admin
- 4 Phone gets CUCM target from DHCP opt 150/TFTP, user enters act. code
- 5 Phone authenticates to CUCM using MIC + activation code
- 6 CUCM updates device config with phone MAC, sends config (+tokens) to phone: phone can now register

cisco Live!

Activation code-based onboarding – MRA flow



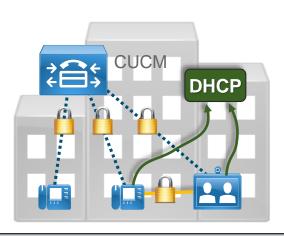
- O Cisco® UCM onboards with GDS, creates MRA target
- 1 Admin creates full device configuration without specifying MAC address
- Admin requests activation code for this device 50% GDS, 50% UCM
- 3 Activation code sent to a user/admin
- User enters activation code, phone gets MRA target from GDS
- Phone authenticates to MRA/UCM using MIC + activation code
- 6 UCM updates device config with phone MAC, sends tokens and config to phone; phone can now register

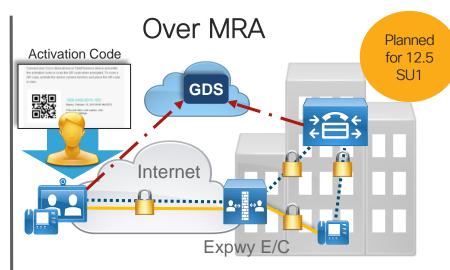
TECCOL-2982

Activation code-based onboarding - Deploying Phones

Available with 12.5

On-premises





Discovery	DHCP option 150	Activation Code
AuthN (bootstrap)	Certificate: MIC (optional)	MIC + Activation Code
AuthN (after)	Certificate: LSC (optional)	OAuth tokens
Provisioning	Auto-reg, Self prov, Manual (with MAC), Manual (without MAC)	Manual (with MAC), Manual (without MAC)

Conferencing



Cisco Meeting Server Capabilities

Video Conferencing Audio Interop Conferencing Web Customization Recording Conferencing



Everyone is invited – audio, video, web

Enjoy One Consistent Meeting Experience From:

- Standards-based video endpoint
- Smartphone or laptop with Cisco Jabber
- Browser with WebRTC
- Cisco Meeting App for PC, Mac or mobile
- Skype for Business





It's Just a Meeting

Video Conferencing



- Feel as though you are really there with high quality video experiences
- Make meetings more enjoyable with consistent experience on room, desktop, and mobile devices
- Make the experience work for you with flexible layouts and controls

Audio Conferencing



- Reduce complexity with a solution that integrates with your existing dial plan
- Give your users more flexibility with:
 - Multiple ways to join
 - Customizable DTMF controls
 - Interactive voice response (IVR)

TECCOL-2982

Web Conferencing

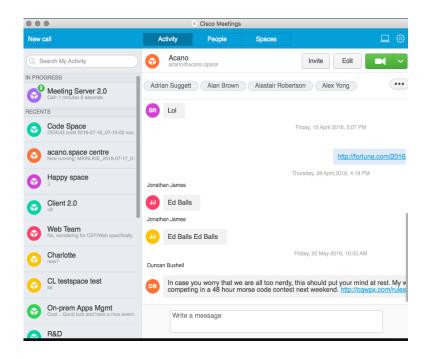


- Attend meetings on the move; join from your browser as user or guest
- Fully participate with audio, video, content sharing, and controls
- No plug-in with WebRTC compatible browsers
- Maintain security with guest access PIN or user login



Meetings

- Virtual rooms for audio, video, and web
- Unlimited spaces: as many as you want, for each project, and team
- Always available
- Follows your workflow
- Connect with any device



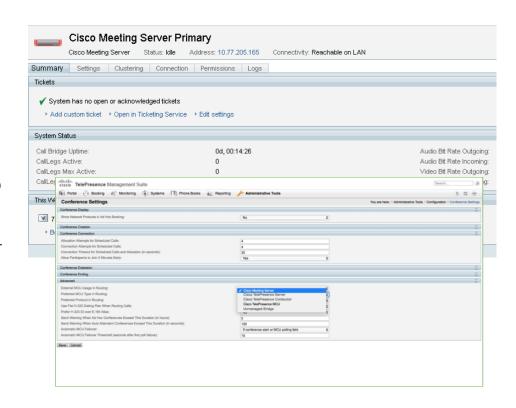
Personal, Scheduled or Ad-hoc Meetings

- Personal meetings:
 - Invite others to your personal meeting using your own join details
 - With Spaces users are in control
- Schedule meetings:
 - Leverage Cisco TelePresence Management Suite (including Microsoft Outlook integration)
 - One-Button-to-Push support
- Ad hoc with UCM
 - Easily escalate your 1:1 calls to include more people



Meeting Scheduling

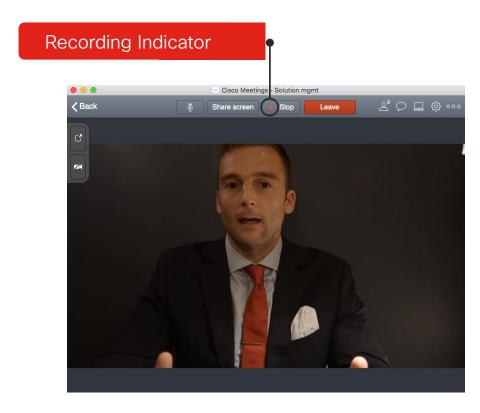
- Cisco TelePresence Management (TMS)
 Suite
- Scheduling using Outlook, HTML Smart Scheduler, by helpdesk, or booking API
- One Button to Push (OBTP) with Cisco video endpoints to easily join meetings
- OBTP with Skype for Business Outlook plugin with Meeting Server Dual Home solution
- Co-exist with TelePresence Server & MCU





Meeting Recording

- Record your meetings for later viewing
- Integrated Record button, or automatic recording on meeting start, or triggered via DTMF; administrator defined
- Store on Network File System (NFS)
- MP4 file format
- Direct Integration with VBrick REV for portal playback, speech to text, and other playback options.

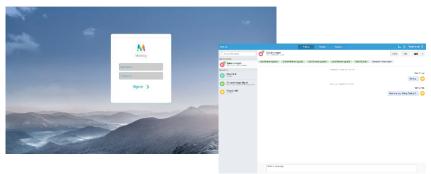




Cisco Meeting App

Enable anyone to join your meeting with the Cisco® Meeting App:

- Fully participate with the benefit of exceptional video quality
- Control your experience with
 - Layout controls
 - Ability to invite others
 - Easily create new Spaces
 - Device mobility and flexibility
 - · Roster list and mute controls
- WebRTC enabled browser support
 - · Easily invite guests to join





Meet the way you want

Flexible In Meeting layouts

Layout Families onePlusN **Active Speaker** Overlay Equal NxN onePlusN and Equal layouts dynamically scale as more participants join onePlus5 onePlus7 onePlus7 Equal 2x2 Equal 3x3 Equal 4x4 Equal 5x5

Customizable Branding



Customize

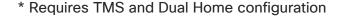
- Backdrop image
- On screen text
- Audio Prompts

Skype for Business Integration

Key Features:

- Seamlessly connects Skype for Business with other video systems
- Skype video, audio, and content sharing
- NEW Schedule conferences from S4B with OBTP*
- Bidirectional content sharing (RDP)
- Direct Federation (including O365)
- Integration Modes
 - Gateway, Spaces, Dual Home





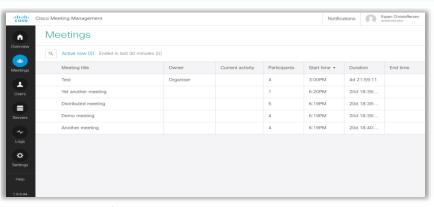


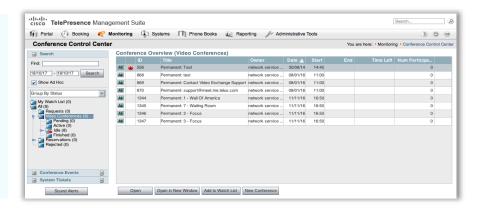
Cisco Video Conferencing Management

New generation Meeting Management

TMS Conference Control Center

- Integrated in TMS
- MCU and TS only
- Windows and Java based



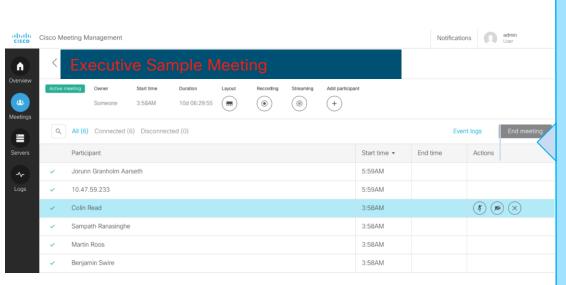


Cisco Meeting Management

- Built on same platform as CMS
- Future proof for adding more services
- CMS only



Cisco Meeting Management 1.1



- Manage active Cisco Meeting Server meetings for white glove operator services
- Future releases to include configuration services and CMS cluster dashboard
- Simplify the deployment with easy to use tools
- Use with Cisco TMS for scheduling and endpoint management
- CMM is included with existing CMS licensing



Cisco Meeting Management

Feature overview

Meeting Management

- See list of active or recent meetings
- Search by meeting title or individual participant
- Pin meeting at top of the list

Cisco Meeting Server management

- See connected Cisco Meeting Servers
- Add and remove Call Bridge nodes or clusters

Notifications and logs

In-conference controls

- Meeting details
- List of participants in a meeting
- Change layout for all participants in a meeting
- Access controls and see details for individual participants
- View and download event log for a meeting
- Start and stop recording and streaming

Cisco Meeting Management Solution Architecture Single Meeting Management deployment

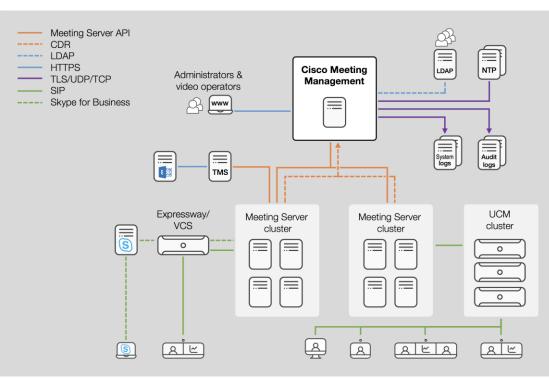
Single instance of Meeting Management can manage deployments from single Call Bridge to multiple Call Bridge Clusters

NTP servers synchronize Meeting Servers and Meeting Management instances

User access is authenticated via LDAP

External syslog server for system and audit logs

Meeting Manager does not require any specific configuration on UCM clusters or Expressway/VCS environment

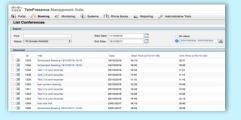




Cisco Meeting Management and TMS

TMS

- Scheduling
- Endpoint Management



Schedule meetings using TMS. OBTP to endpoints.

CMM

 White glove tool (Meeting Manager)



Monitor and manage ongoing meetings using Meeting Management.

NOTE: Information is not exchanged between Meeting Management and TMS in CMM 1.0. Planned for future release.



Applications



Unity Connection



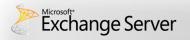
Security and Compliance

- Secure Messaging
 - No chance of forwarding a secure voice message
 - Securely streamed from Connection appliance (Message never leaves Connection)
- Secure Delete
- SE Linux enabled
- Disaster Recovery with Full data backup and restore
- Federal Information Processing Standards (FIPS)
 - Version 11.5(1) Incorporates FIPS Compliant Libraries
- Joint Interoperability Test Command (JITC) certification
 - Version 11.5 (1) JITC Certified and on APL



Interoperability

- LDAP user synchronization
- Single Inbox
- Partner-based Integrations
 - · Google Mail, Domino, etc...
- PIMG/TIMG Integrations
- VPIM Support
- Multiple phone systems supported simultaneously
 - Cisco UCM-SME allows for centralized deployment
- Rest-Based APIs allow for custom development













Integrations

- Supports Exchange 2016, 2010, 2007 and 2003
- Microsoft BPOS-D and Office 365 (8000 supported 11.5.(1))
- Google Mail via Donoma Software
- Domino integrations
 - Intelligent Notifications (Unity Connection 9.0 and later)
 - Donoma Unify (Partner Solution)
 - Esnatech CloudLink (Partner Solution)





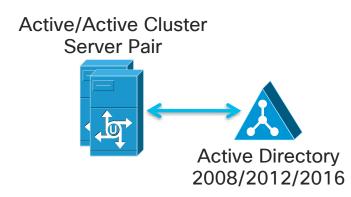






User Directory Synchronization/Authentication

- Synchronization with users in Active Directory 2008/2012/2016 11.5(1), 12.X
 - No schema extensions necessary
 - One-way synch of user data from LDAP (read only)
 - Distribution Lists and Contacts not supported
- Up to 20K users can be synchronized and authenticated per server/cluster
- Filters supported per synchronization agreement
- Directory integrated users, standalone users, and CUCM AXL users can co-exist on Unity Connection server
- Standalone and AXL users can be converted to directory integrated users
- LDAP/AD Synchronization is NOT required





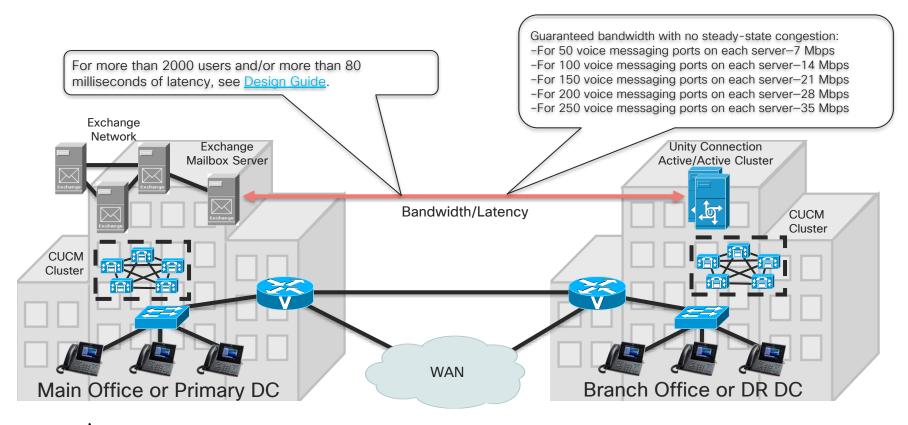


Unity Connection Scale

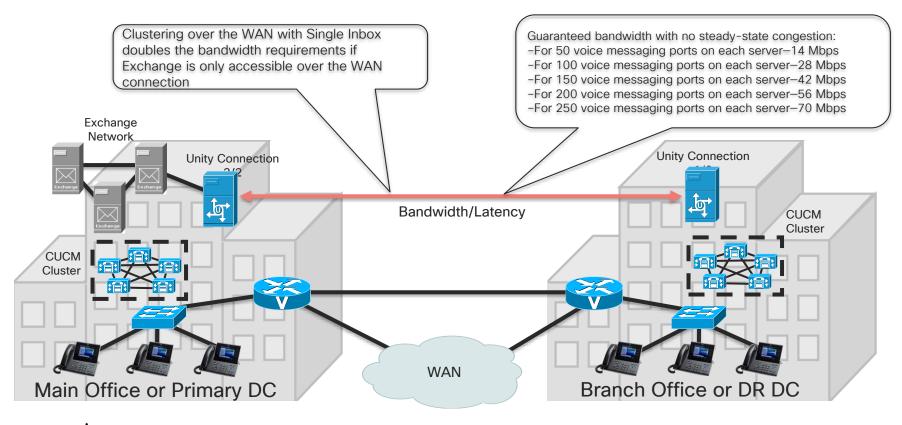
- Up to 20,000 users and 250 ports per server
- Active/Active Redundancy over LAN/MAN/WAN (up to 500 ports)
- 100,000 users across 20 digitally networked nodes
- VPIM Networking (100 locations, 100,000 users/contacts)
 - Blind addressing doesn't count against contacts



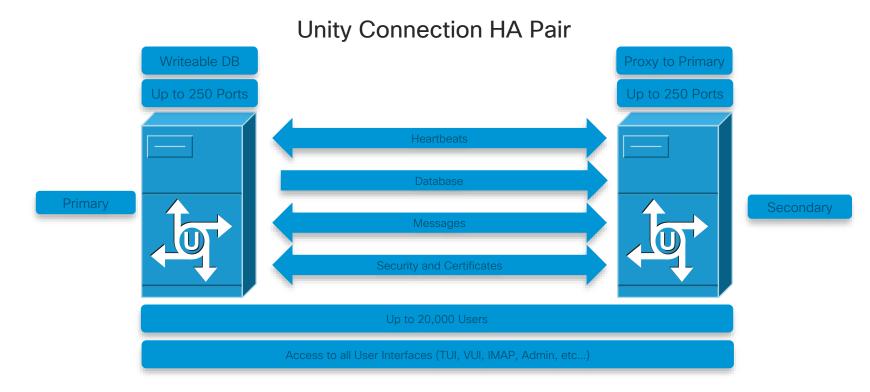
Single Inbox - Design Considerations



Single Inbox - Design Considerations



High Availability - Failure





Solutions Plus



Imagicle



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imagicle°

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Cisco Paging Server



What is Cisco Paging Server?

- With every new CUCM and BE6000/7000 starting with 9.1, the Informacast Paging Software OVA is included with all shipments.
- Existing customers on CUCM 8.X and newer will be able to download Informacast Paging Features from the CCO access with a valid ESW Contract.
- ISO includes OVA for VM Deployment and production documentation as well.
- Fully supported as a Co-Resident virtual server alongside other Cisco approved voice applications.
- Limited functionality included in the basic licensing model. Advanced notification licensing can be purchased to un-lock the full features of Informacast.
- 60-Day trial of Advanced notification is available to determine if it's a need for the environment.



What is the difference between Basic and Advanced?

Basic Paging - Free Not Licensed

- Point to Point and Group Live Audio Paging to/from Cisco Phones
- Unlimited Group/Zones of endpoints as configured by admin
- Maximum of 50 endpoint devices per group
- Paging between sites is supported (Multicast over WAN)

What is the difference between Basic and Advanced?

Advanced Notification - Optional and Licensed

- Pre-Recorded/Scheduled Broadcasts (bells, shift changes)
- Notification to Jabber (XMPP)
- Notification to Social Media
- Communication with mobile and remote users
- Triggered notification to/from other systems M2M input/output (panic buttons, door locks, lights, etc.)
- Integration with existing overhead paging systems
- Text and Audio to Cisco IP Phones and other endpoints
- Broadcast to IP Speakers



What is the difference between Basic and Advanced?

Advanced Notification - Additional Components

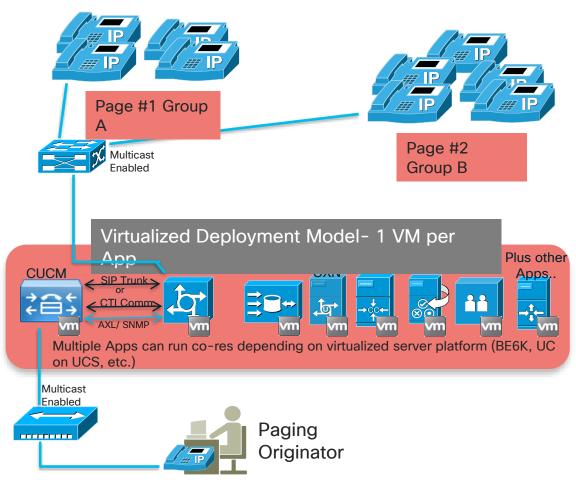
- Legacy paging integration with existing overhead paging systems
- 911/Emergency call monitoring/alerting/recording
- Weather and CAP Based Alerts
- Dynamically-triggered conference calls
- Notifications to desktop and Digital Signage



How does it work?

Integration with CUCM

- Two methods of communication - CTI and SIP
- Services required for backend connectivity -SNMP and AXI
- Phones require the following features - HTTP (web server) and RTP (Multicast)



Advanced Notification



On-Premise - IP Speakers

Direct Support in Informacast



- Speakers must be supported in Informacast
 - Manufacturers included Atlas Sound, Advanced Network Devices, Cyberdata, and Valcom, Algo.
- There are many styles/models of Speakers
 - Most are POE based, Single connection and require Class 3 Power over Ethernet
 - Some are One-Way, Some are Two-Way and some support dual registration with CUCM as a SIP endpoint for direct calling (Enhanced User connect license needed)
 - · Outdoor, Indoor, Indoor with Display, Indoor with Display and Strobe, Ceiling Tile
 - More Information: <u>IP Speaker Information</u>
- Connect to existing Amplifiers with a Zone controller, which provides a line level output to existing over head speakers.

Remote Access

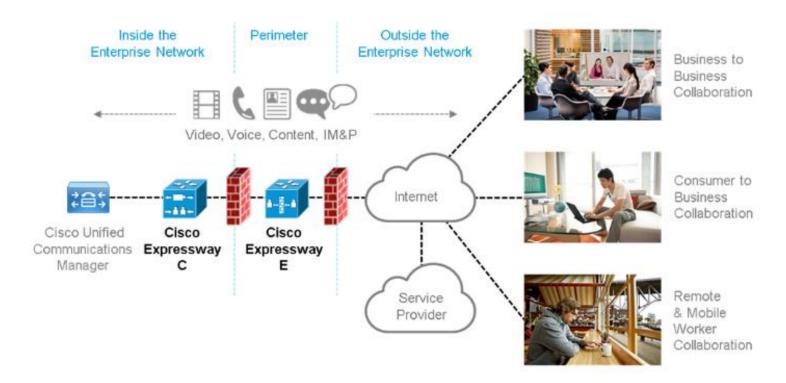


Expressway
Solution Overview



The Collaboration Landscape

Enabling Connection of Any Users to Any Services Through Any Network





Cisco Expressway



Expressway Compute Platform Options

Specs Based Virtual Machine Support						
OVA Size	vCPU	Reserved RAM	Disk Space NIC(s)			
Small	2 x 1.8 GHz	4GB	132GB	1Gb		
Medium	2 x 2.4 GHz	6GB	132GB	1Gb		
Large	8 x 3.2 GHz	8GB	132GB	1Gb		





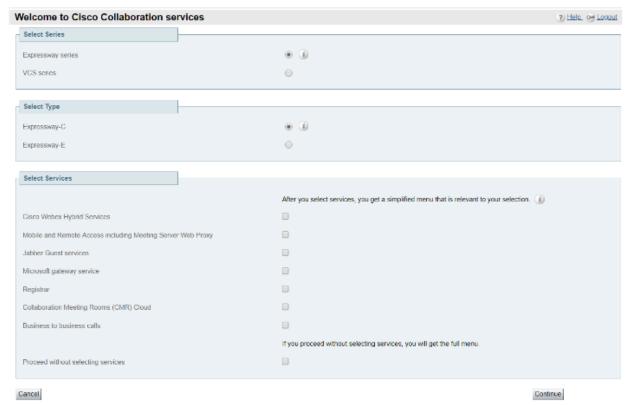
Expressway 12.5 Scalability

	Server			Cluster		
Platform	Registrations	Video Calls	Audio Only Calls	Registrations	Video Calls	Audio Only Calls
CE1200	5,000	500	1000	20,000	2,000	4,000
Large OVA, CE1100	2,500	500	1,000	10,000	2,000	4,000
Medium OVA	2,500	100	200	10,000	400	800
Small OVA (BE6000 M5)	2,000	75	150	2,000	75	150

X8.10 introduces large-scale capacity on Large OVA and EXPWY1200 with only 1Gbps NIC



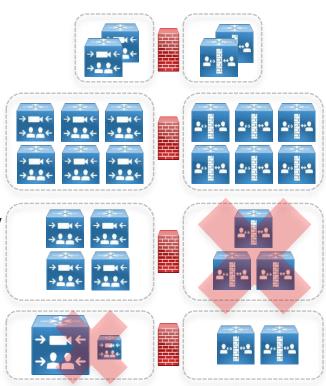
Service Setup Wizard





Expressway Clustering, 4+2

- Cluster up to 6 Expressways for scale and redundancy
- Clustering latency up to 80ms RTT
- Expressway E and C node types cannot be mixed in the same cluster
- Deploy equal number of peers in Expressway C and E clusters (this applies to most Expressway deployments but is not critical if Expressway is handling local registrations)
- Deploy same OVA sizes or appliances throughout cluster
- Customers can deploy multiple clusters for the same domain



Expressway Licensing

Business to Business Calls

Firewall Traversal Calls consume 1 x RMS on Expressway E (includes MSFT B2B calls)

Interoperability Gateway Calls

i.e. intradomain MSFT interop calls, consume 1 RMS on Expressway C Gateway

Device Registration

Desktop and Room registrations provided by UCI

Registered Calls (no RMS required)

Calls between endpoints registered to Cisco Call control services
Calls to Cisco conferencing infrastructure **or cloud services**Cisco Meeting Server WebRTC



Expressway License & Resource Usage

 Calls from MRA endpoints or endpoints registered locally to Expressway are classified as Registered calls

 Calls to/from Webex cloud are classified as CMR Cloud calls

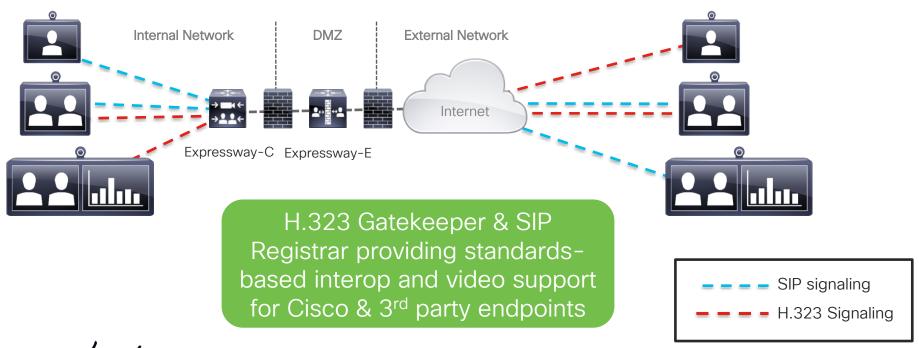
 "Registered" & "Cloud" calls do not consume Rich Media Session licenses, but do count against the overall system capacity

		Total	10.9.8.107	10.9.8.1
Registered calls	Current video	1	1	0
	Current audio (SIP)	1	0	1
	Peak video	5	3	2
	Peak audio (SIP)	5	3	2
	Since last restart video	120	88	32
	Since last restart audio (SIP)	43	30	13
CMR Cloud calls	Current video	0	0	0
	Current audio (SIP)	0	0	0
	Peak video	1	1	0
	Peak audio (SIP)	0	0	0
	Since last restart video	1	1	0
	Since last restart audio (SIP)	0	0	0
Rich media session calls	Current video	0	0	0
	Current audio (SIP)	0	0	0
	Peak video	0	0	0
	Peak audio (SIP)	0	0	0
	Since last restart video	0	0	0
	Since last restart audio (SIP)	0	0	0
Monitored resource usage	Current		1%	1%



Expressway Local Registration Support

SIP & H.323 Video Call Control



Expressway-E Local Registration Support

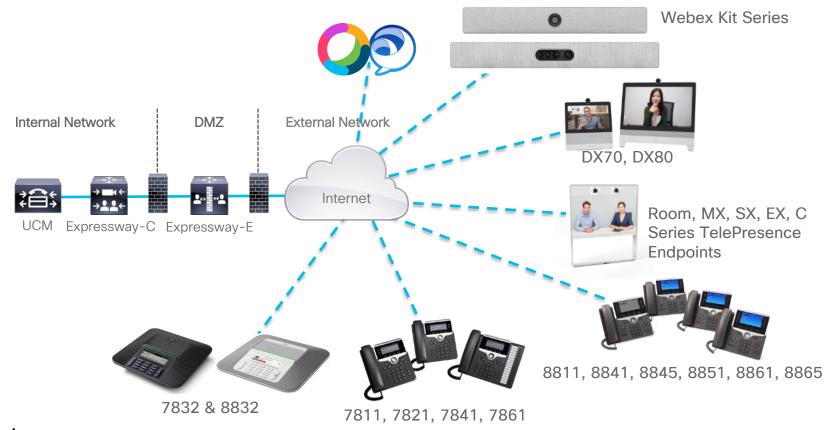
- Beginning with X8.11 <u>Expressway-E</u> will support local SIP and H.323 video registration
 - Expressway-E no longer required to proxy SIP registrations
 - Allows for remote H.323 registrations
- Local SIP and H.323 registrations allowed on <u>Expressway-C</u> since X8.9
- UCL Enhanced License enables SIP Desktop Endpoints (DX70/80, EX60/90)
- TP Room System License required for all other systems including 3rd party and H.323
- Same option keys (Room System, Desktop System) used on both Expressway C & E



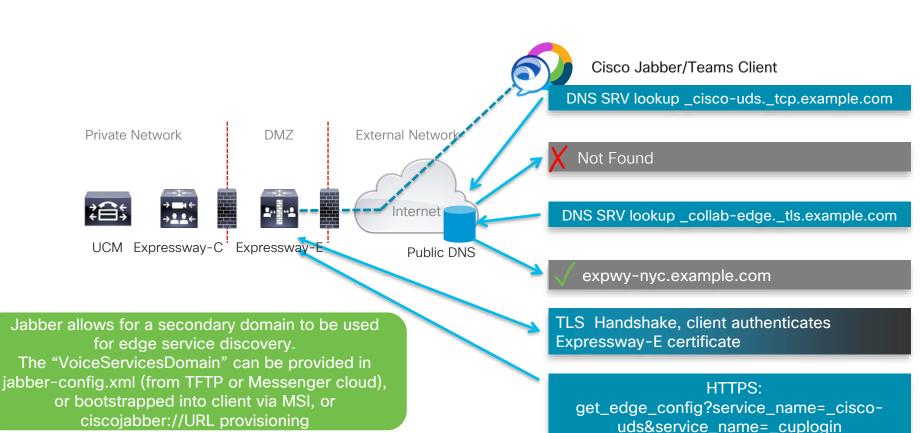
Expressway
Mobile Remote
Access (MRA)



MRA Client & Endpoint Support



Expressway & Service Discovery



TECCOL-2982

cisco Live!

Split DNS SRV Record Requirements

- collab-edge record needs to be available in public DNS
- Multiple SRV records (and Expressway-E hosts) should be deployed for clusters
- cisco-uds record needs to be available only in internal DNS

<u>collab-edge._tls.example.com. SRV 10 10 8443 expwy1.example.com.</u> _collab-edge._tls.example.com. SRV 10 10 8443 expwy2.example.com.

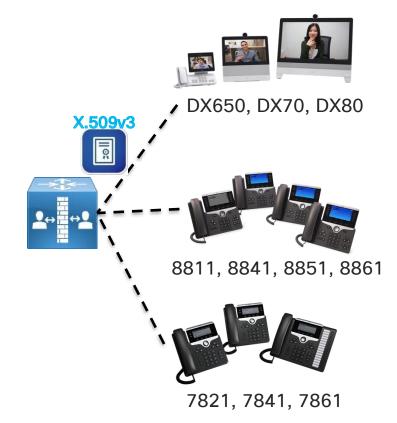
_cisco-uds._tcp.example.com. SRV 10 10 8443 ucm1.example.com. _cisco-uds._tcp.example.com. SRV 10 10 8443 ucm2.example.com.



Expressway-E certificate requirements

DX, 78XX, 88XX specific requirements

- Trust model based on broadly trusted Public Certificate Authorities
- Endpoint firmware includes 135 trusted public root CA certificates
- No option to import and trust other root CA certificates on these endpoints
- Expressway-E certificate needs to be signed by trusted public CA



Expressway 12.5: Let's Encrypt

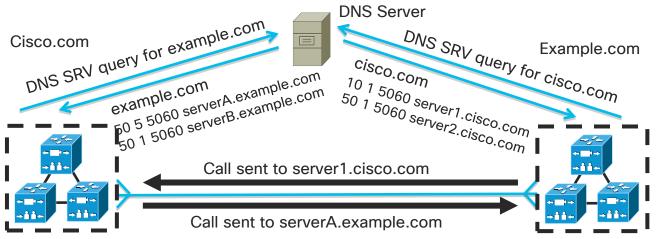
- Let's Encrypt is a free, automated, and open Certificate Authority
- Providing X.509 certificates for TLS encryption at no charge
- Includes an automated process designed to overcome manual creation, validation, signing, installation, and renewal of certificate
- ACME protocol (Automated Certificate Management Environment)
- More details at <u>www.letsencrypt.org</u>
- Let's Encrypt signed certs are compatible with all MRA endpoints
- Compatible with both Expressway server and domain certificates



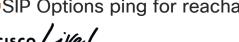
Expressway
Business to
Business (B2B)

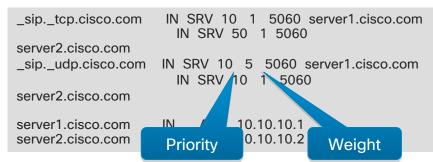


DNS Based Domain Services

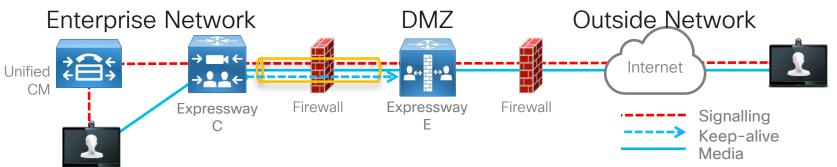


- •SIP, H.323 service
- TCP or UDP protocol
- •Multiple servers in record provide redundancy
- •DNS SRV records use priority, weight for load balancing
- •DNS server needs to be highly available
- SIP Options ping for reachability





Expressway Firewall Traversal Basics (Inbound)

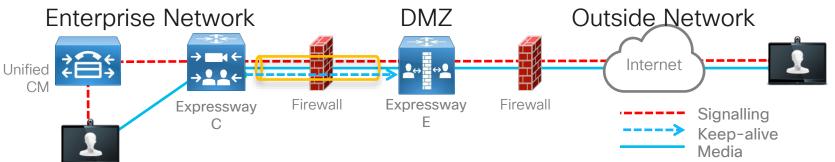


- 1. Expressway-E is the traversal server installed in DMZ. Expressway-C is the traversal client installed inside the enterprise network
- 2. Expressway-C initiates traversal connections outbound through the firewall to specific ports on Expressway-E with secure login credentials
- 3. Once the connection has been established, Expressway-C sends keep-alive packets to Expressway-E to maintain the connection
- 4. When Expressway-E receives an incoming call, it issues an incoming call request to Expressway-C
- 5. Expressway-C then routes the call to Unified CM to reach the called user or endpoint

 The call is established and media traverses the firewall securely over an existing traversal connection



Expressway Firewall Traversal Basics (Outbound)



- 1. Expressway-E is the traversal server installed in DMZ. Expressway-C is the traversal client installed inside the enterprise network
- 2. Expressway-C initiates traversal connections outbound through the firewall to specific ports on Expressway-E with secure login credentials
- 3. Once the connection has been established, Expressway-C sends keep-alive packets to Expressway-E to maintain the connection
- 4. When an endpoint behind **Unified CM** places a SIP URI call to a domain unknown by the **Unified CM** dial plan, the call is routed to the **Expressway-C**
- 5. Expressway-C then routes the call to Expressway-E to reach the called user or endpoint
 The call is established and media traverses the firewall securely over an existing traversal connection



SIP URI Dialing

user @ example.org

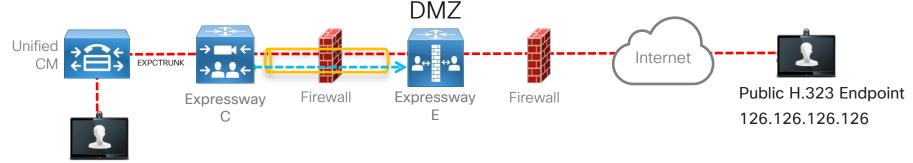
left-hand-side (LHS) right-hand-side (RHS)



DN:\+14085552001 srogers@videolab.com

- User srogers dials the SIP URI chuck@cisco.com
 - a. Is the LHS numeric? NO
 - b. Does the URI match a pattern in the CSS or URI table or one in ILS? NO
 - c. Does the RHS match a specific SIP Route Pattern? NO
- Route the call based on RHS to Expressway C.
- 3. Expressway C searches its zones for anything that matches chuck@cisco.com, finds nothing, and routes the call over the Traversal Zone to the Expressway E.
- 4. Expressway E searches its zones for anything that matches chuck@cisco.com, finds nothing, and routes the call out the DNS zone.

IP Address Dialing Use Case: Business to Business

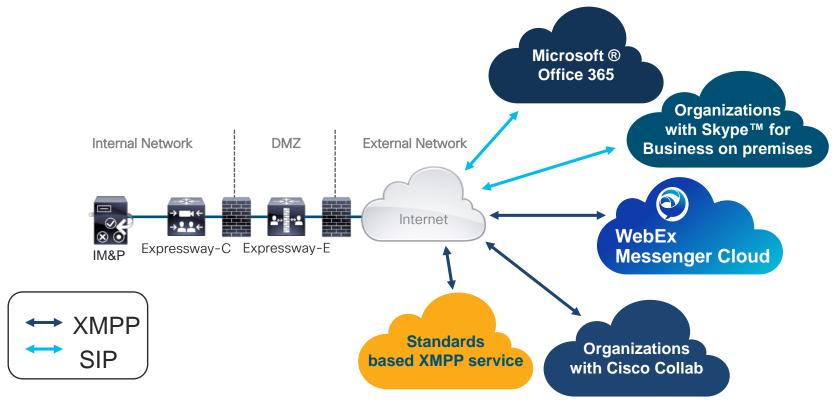


DN:\+14085552001 srogers@videolab.com

- User srogers dials the SIP URI <u>126.126.126.126@ip</u> (RHS can be any keyword)
- 2. Matches Unified CM SIP Route Pattern of * and routes to EXPCTRUNK device.
- 3. Expressway C matches the URI with pre-search transform that strips the RHS @ip and forwards the call to Expressway E.
- Expressway E matches identifies the destination as an IP address and forwards to the IP Address Zone and sends the call to the destination IP of 126.126.126.126.

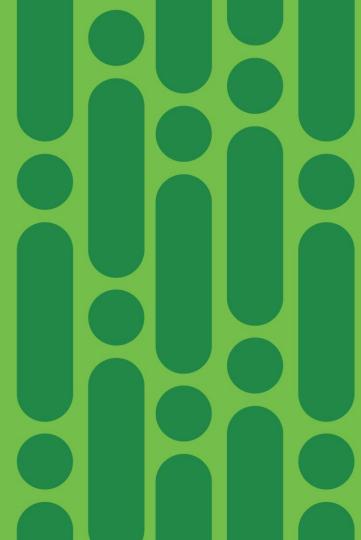
Jabber IM&P Federation with Expressway

Now extending to organizations using Microsoft



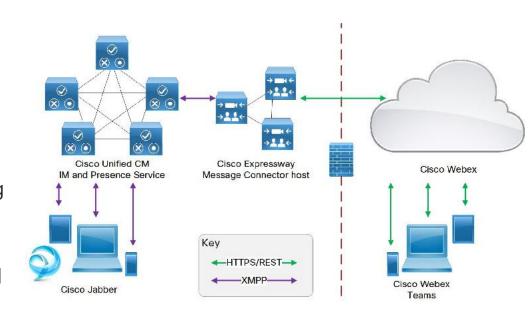


Hybrid Messaging with Webex Teams



Cisco Webex Teams and IMP Hybrid Messaging

- Hybrid Service to enable Messaging between on premise IMP server and Webex Teams Cloud Messaging.
- Allows Jabber users to message with Teams Users in a one-to-one option.
- Meant as a bridge solution for migration, not really meant for a long term deployment (feature gaps)
- With Jabber Teams mode and the Webex client, long term use or need for premise is not needed
- You will lose your XMPP federation functionality



Prime
Collaboration
and
Telepresence
Management Suite



Cisco Prime Collaboration

- Cisco Prime Collaboration Provisioning
 - Single point of Move/Add/Change Mechanism
 - LDAP is Required
 - Single pane of Glass and Multi-Levels of Administration (Help-Desk)
 - Template and Service Area Based
 - · Start with Prime Provisioning, overlay on existing system can take more time
 - Auto-Provisioning from LDAP, User Portal
 - Virtual Server, with mutltiple sizing based. Linux based deployments



Cisco Prime Collaboration

- Cisco Prime Collaboration Assurance and Analytics
- Prime Collaboration Assurance
 - The former "Operations Manager and Service Monitor"
 - Assurance for one Cluster is included with all CUCM 10.X and above deployments
 - Analytics is a cost per "device"
 - Red light/Green light view of the voice and network portion
 - Great intermediary between the CUCM and a overall management platform.
- Prime Collaboration Analytics
 - Call quality and MOS score tool
 - Can actively provide analytics during call (1040 probe required)



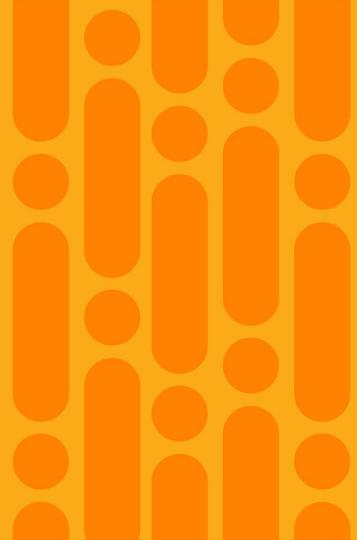
Telepresence Management Suite

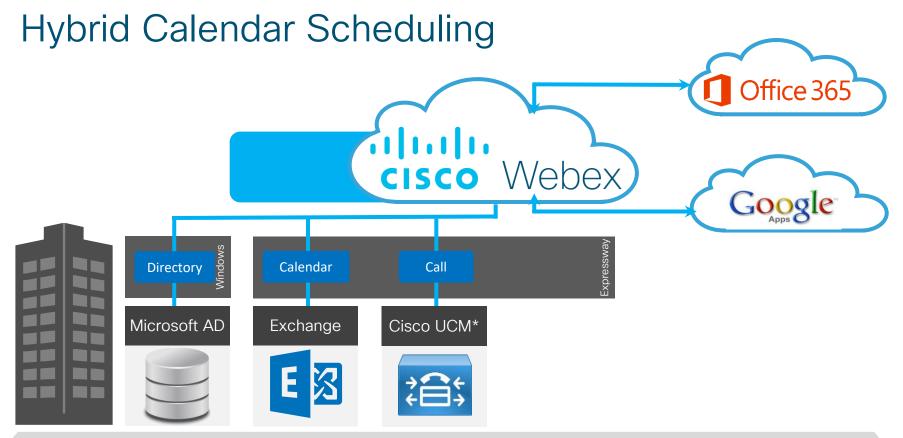
Cisco TMS

- Video Management and booking engine
- Manage VCS/Expressway Health, Call Records, Issue management
- Manage Endpoints, booking of Endpoints, booking of Bridging resources
- Live View of conferencing resources, ability to add/remove users in a live conference
- Multiple Extensions to enable Exchange resources booking, API integration, Provisioning for Endpoints/Jabber for Telepresence (MOVI). Third party bridging and endpoint support available as well (check the compatibility matrix)
- Start with Prime Provisioning, overlay on existing system can take more time
- Auto-Provisioning from LDAP, User Portal for simple scheduling.
- WebEx CMR and CMR-Hybrid booking mechanism



Hybrid Architecture

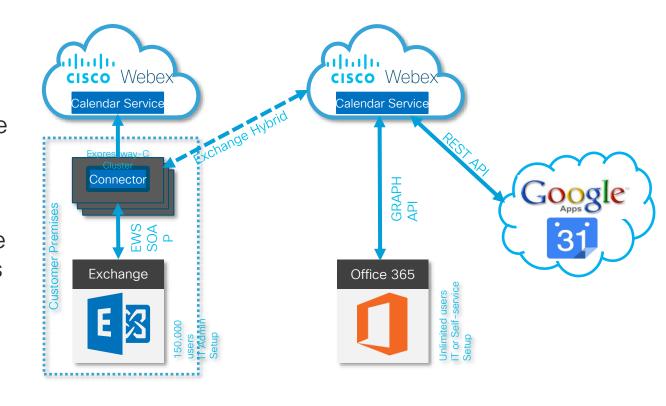




*Includes Business Edition or HCS

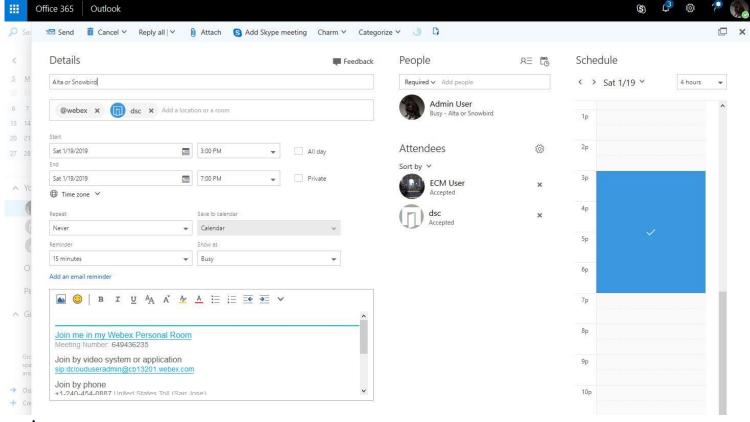
Calendar Integrations

- @webex or @space in a calendar invite
- @webex backfills the users PMR information into the invite
- @meet will utilize the Cisco Webex Teams space information when populating the invite, or create a new space



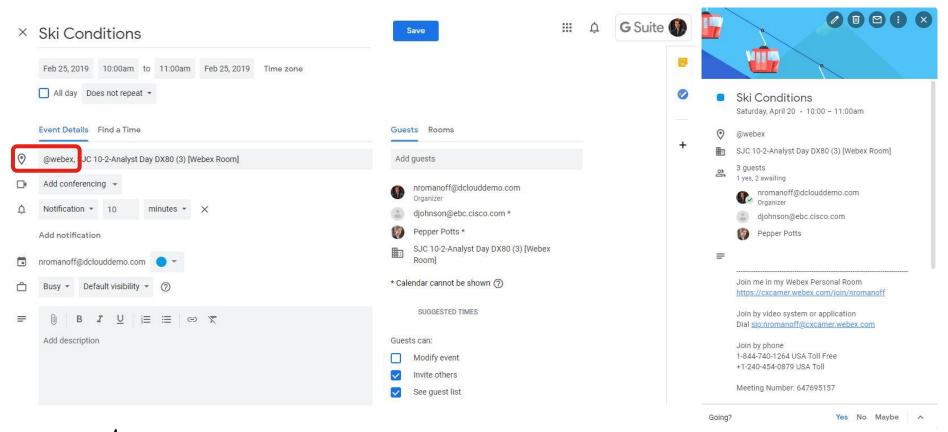


O365 - Cloud Calendar Connector

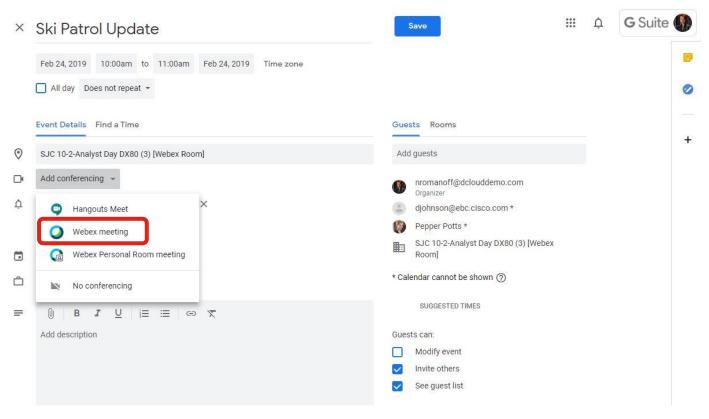




Google G-Suite - Calendar Service Scheduling

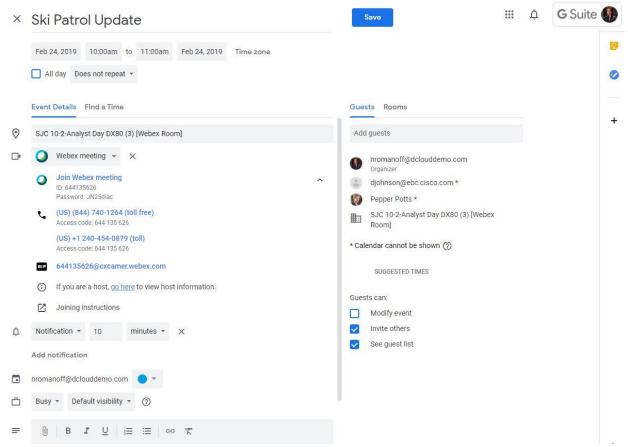


Google G-Suite - Webex Meetings Scheduling

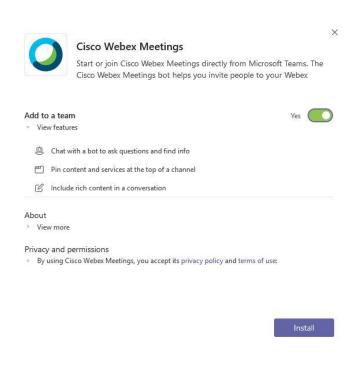


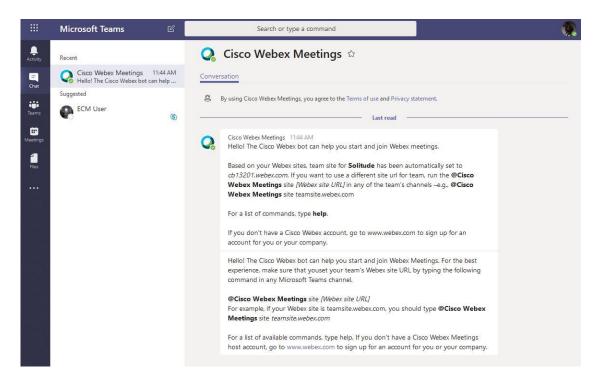


Google G-Suite - Webex Meetings Scheduled



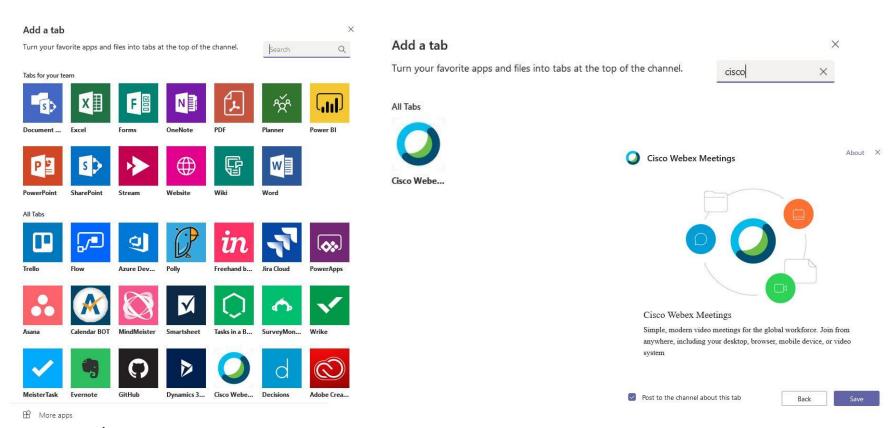
Microsoft Teams - Webex Meetings Bot



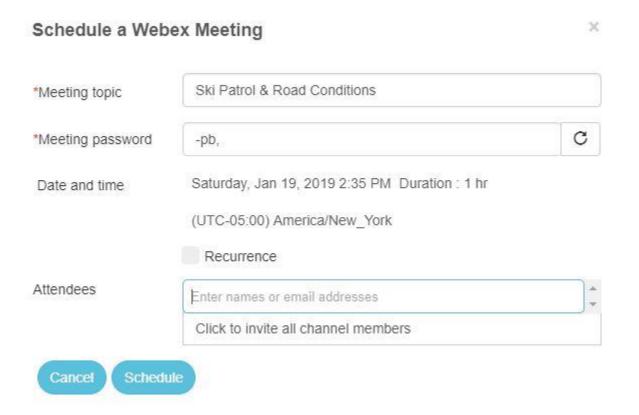




Microsoft Teams - Webex Meetings Tab Install

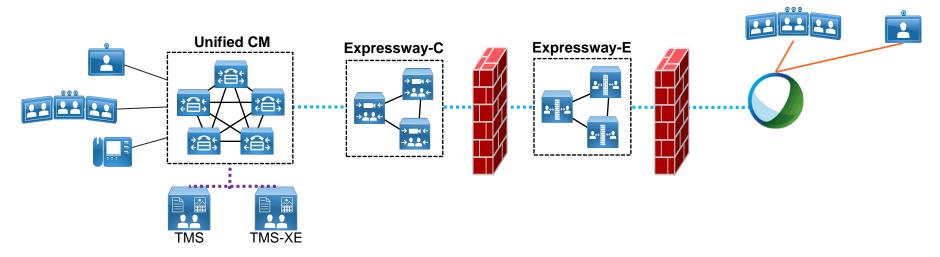


Microsoft Teams - Webex Meetings Scheduling





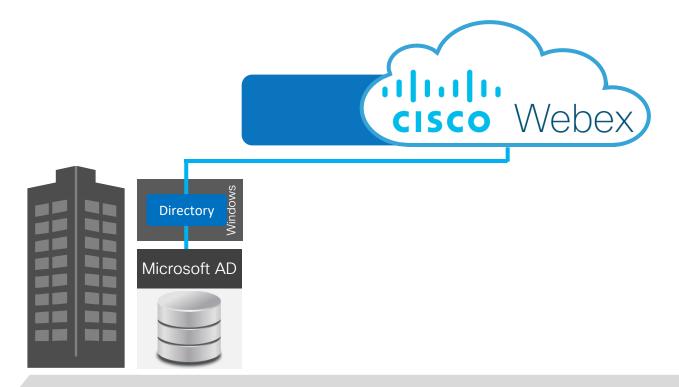
Hybrid Scheduling



- Device must be managed by Cisco TelePresence Management Suite:
 - Set up Cisco TMS 15.0 and Cisco TMSXE 5.0 or higher with Microsoft Exchange integration - <u>Cisco Collaboration Meeting Rooms (CMR) Hybrid Configuration</u> <u>Guide (TMS 15.0 - WebEx Meeting Center WBS30)</u>
- To make conference rooms schedulable in Microsoft Outlook/Exchange, configure them in XE as if you were using on-premises conferencing - <u>Cisco TelePresence</u> <u>Management Suite Extension for Microsoft Exchange Administration Guide</u>.

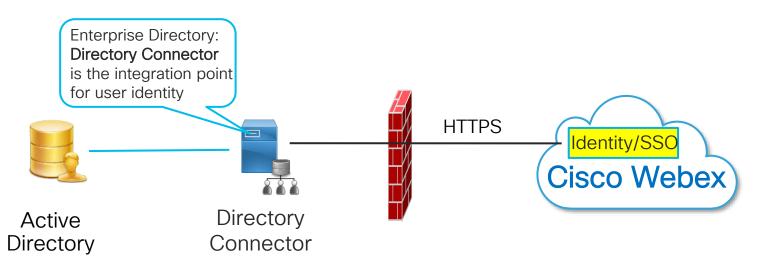


On-Premises Directory Synchronization





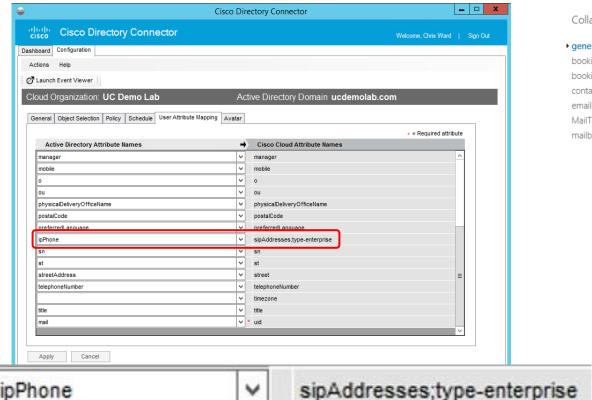
Identity integration for Users & Devices



- > Directory connector integrates with AD to retrieve user information to sync with the identity service, and specifies the Active Directory synchronization agreement and attribute mappings
- ➤ Customer installs Directory Connector in its network on a Windows Domain server (Windows Server 2003, 2008 R2, 2012, 2012 R2, 2016) with administrative user privilege
- ➤ Directory Connector supports Single Forest, Multi-Domain and Multi-Forest, Multi-Domain



Custom Attribute Mapping



Requires Directory Connector 3.0

Collab Summit 3

booking delegates booking options contact information email address MailTip mailbox delegation



Collab Summit 3

general booking delegates booking options contact information

email address
 MailTip

mailbox delegation

default reply address is displayed in bold. To change the default reply address, select the email address that you want to set as the default, and then double-click to edit it.

Email address:

Each email address type has one default reply address. The

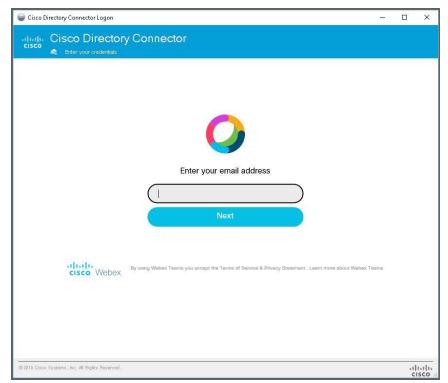
+ / −

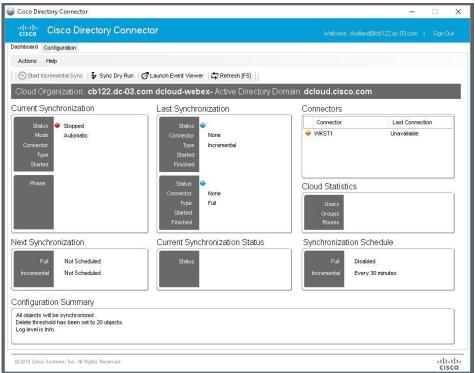
TYPE EMAIL ADDRESS

SMTP 5915@ucdemolab.com

Cisco Directory (DirSync) Connector

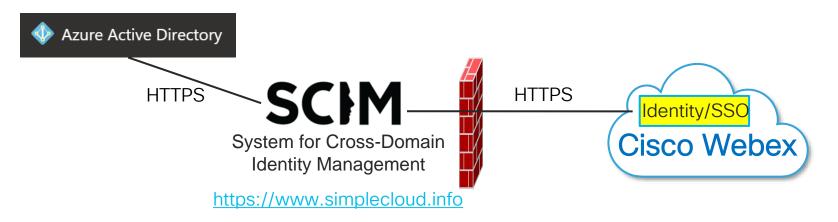








Cloud Identity integration for Users & Devices



- > Simplified user provisioning and lifecycle management
- ➤ Enterprise Authentication via enterprise standard SAML IDP or industry leading IDaaS vendors
- ➤ API Security OAuth 2.0 provides API security for integration with third party SaaS providers



Cloud Architecture



Cisco cloud collaboration strategy



One cloud platform



One core experience



One brand name





Cloud Architecture Components

- Control Hub
- Messaging
- Meetings
- Security
- Calling

Control Hub



Management Portal - evolution

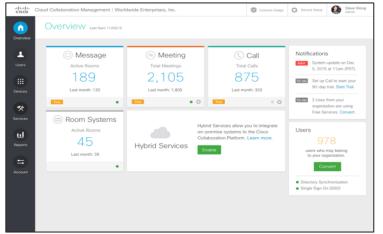






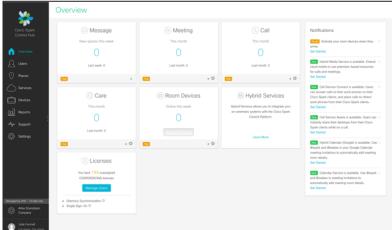
Circa 2015

Cisco Cloud Collaboration Management



Aug 2017







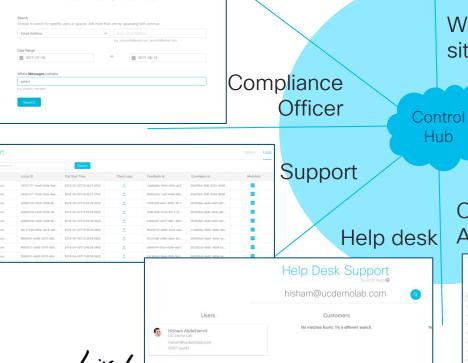
April 2018 ()

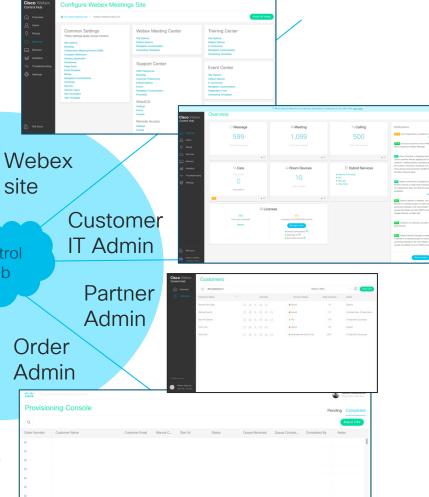


Webex Control Hub

Cisco Webex Control Hub

Oh the wonderful things Control Hub can do!



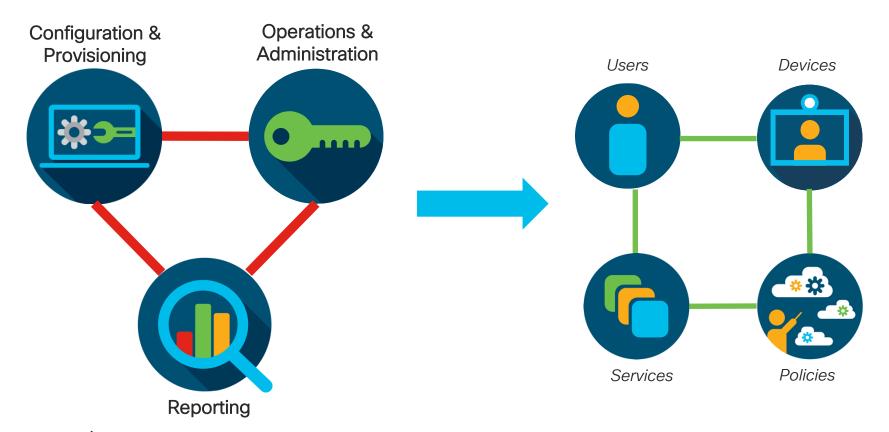


User roles in Control Hub



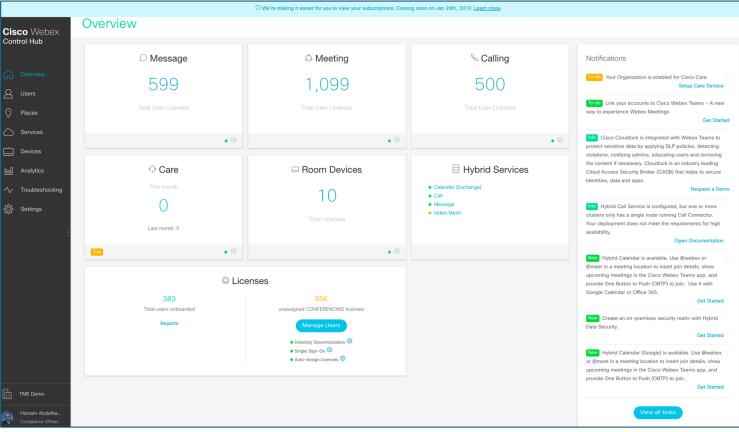
Access	Full Admin	Read-only Admin	Support Admin	User & Device Admin	Device Admin	Compliance Officer
Activity reports						€
Analytics and Diagnostics	₹	₹	V			
Device Management	₹	Read-only		₹	€	
User Management	€	Read-only		€	Hisham Abdelhamid hisham@ucdemolab.com User > Roles and Security Vuser Management Company Policy & Templates Analytics & Reports Analytics & Reports Some administrator privileges Support Administrator Vuser Management Company Policy & Templates Analytics & Reports Vuser Management Company Policy & Templates Analytics & Reports Vuser Management Company Policy & Templates Vanalytics & Reports Vuser Management Company Policy & Templates Vanalytics & Reports Vuser Management Company Policy & Templates Vanalytics & Reports Vuser Management Vuser Management Vouer	
Assign Roles	₩	Read-only		 ✓		
Support metrics & Notifications	€	Read-only	€			
Licenses & Upgrades	€	Read-only	€	€		
Company Policy & Templates	₹	Read-only			evice Administra <mark>tor ()</mark> inistrator ()	
https://collab	orationhelp.cis	: sco.com/article/en-	us/fs78p5#CMG	<u>'</u> iT_TK_T904D062_00	Compliance Compliance Office	r ①

Webex Control Hub basic functions



Overview page, Customer Admin







Control Hub "Overview" landing page

- This is the first menu an admin encounters upon login to Control Hub
 - · Viewable by full and read-only admins
- Provides a summary of the Webex Services provisioned for this organization
 - Cards for each Service: Message, Meeting, Calling, Care, Room Devices, and hybrid services
 - Hybrid Services configured and a quick summary of their status
 - License consumption/overage, on-boarded users summary
 - Notifications of actions required
 - Tasks





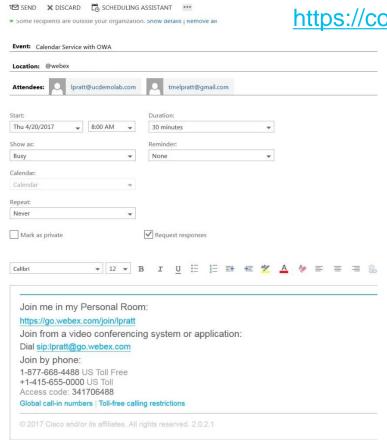
Cloud Connectors



Calendar Connector Office 365 cisco Webex Directory Call Microsoft AD Cisco UCM*



Calendaring Scheduling Integration with @webex



https://collaborationhelp.cisco.com/article/en-us/nwtulzs

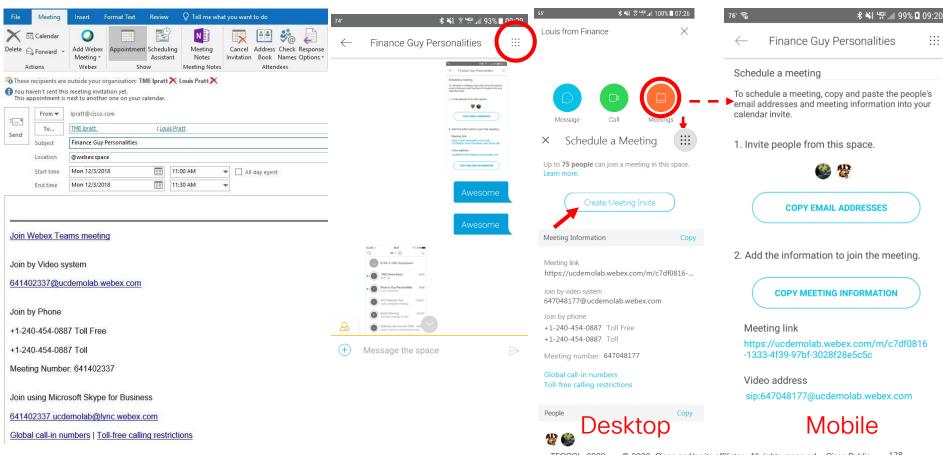
Users can use either @webex or @meet in a calendar scheduling invite, not both.

@webex will backfill the users personal room information into the calendar invite (seen here)

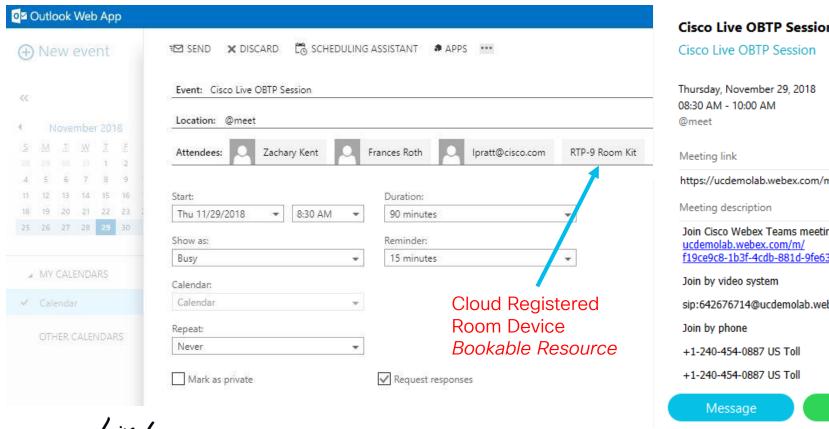
@meet or will utilize the Cisco Webex Teams space information when populating a calendar invite, or create a new space (next page).

There are also modifiers :space and :myroom

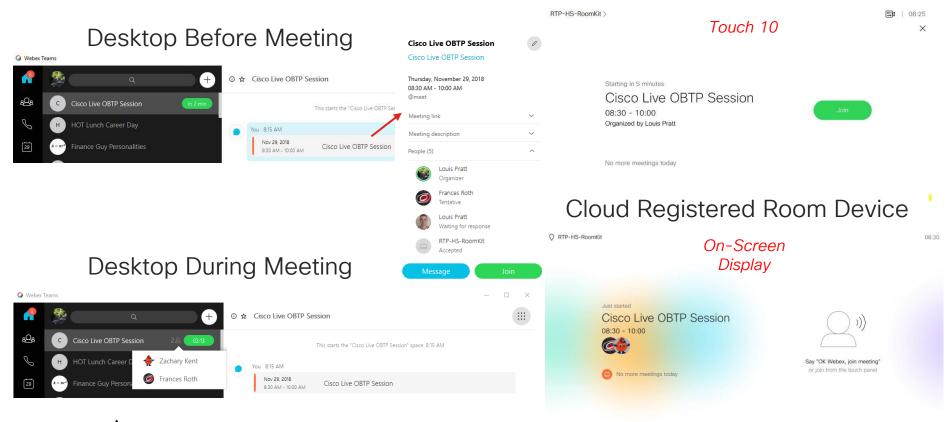
Calendaring Scheduling Integration with @meet



OBTP User Meeting Scheduling Experience



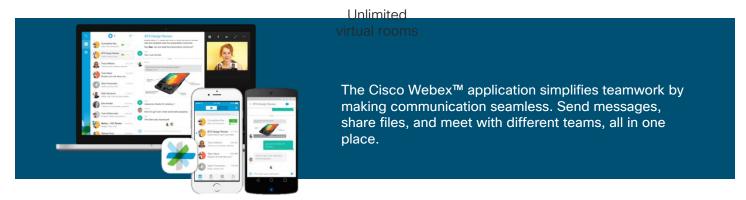
OBTP User Meeting Join Experience



Messaging



Webex Teams Messaging













Persistent and secure messaging and file sharing

with screen sharing

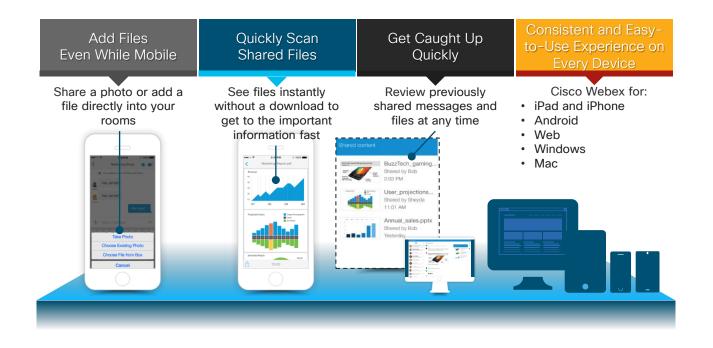
Face-to-face meetings Superior business-class experience



Security and Compliance Differentiators

- Data Privacy with End to End Encryption
- 2 Hybrid Data Service (HDS)
- 3 Man in the Middle Attack Protection
- 4 External Collaboration + DLP Policy Enforcement
- **(5)** Legal Hold
- 6 Mobile Security
- 7 One Drive/Sharepoint Online Integration

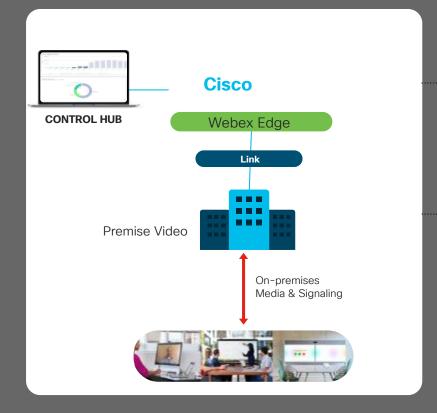
Webex Teams: Get It Done Faster





Single Platform Advantage: New Webex Edge for Devices Webex Control Hub Tools for Any Device Deployment Model





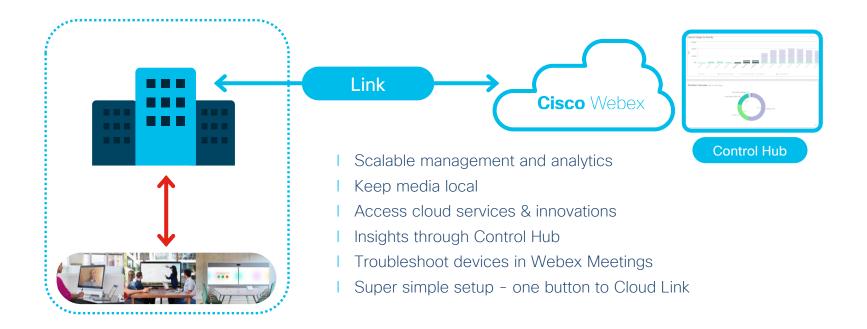
Video innovation

- Extend cloud experiences to on-prem registered devices
- Keep media local, while gaining access to cloud services

Single Platform Advantage

- Control Hub: Management, analytics & diagnostics
 - Simple setup one button to link to Webex Cloud
 - Troubleshoot devices in Webex Meetings
- Enhanced Security
- Advanced cognitive intelligence services Access cloud services & innovations

Introducing Webex Edge For Devices Cloud Experiences to our Premise Customers



Webex Edge for Devices



Business-critical calling stays on-prem

Hybrid deployment and customer migration at their pace

«One-click Migration»

Simplified Single Plane Of Glass

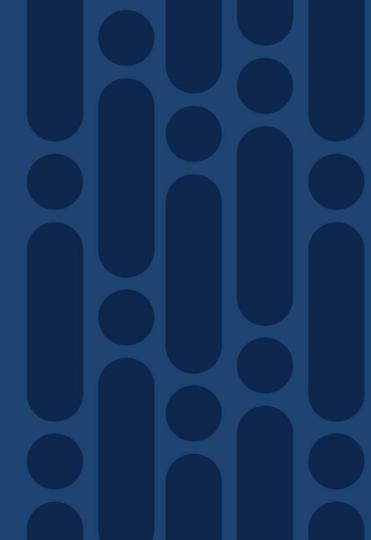






Meetings







Webex Meetings



Designed to deliver the best meeting experience in the industry

Video-first user interface

New desktop, mobile and web apps

Integrations

The Cisco Webex Meetings experience

Video-first

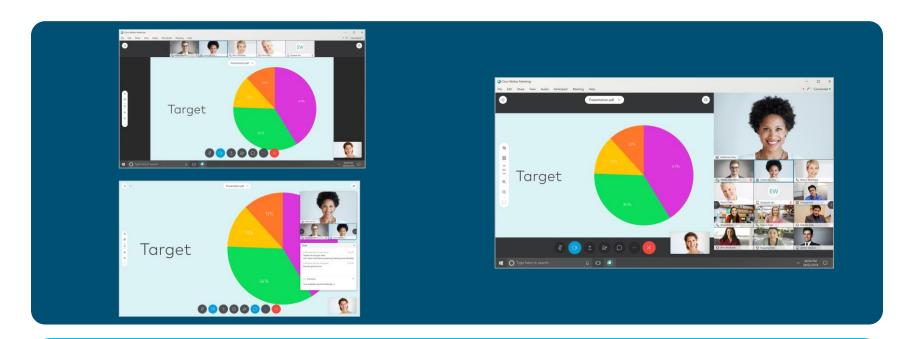


Autonomy to customize video layout



The Cisco Webex Meetings experience

Content sharing

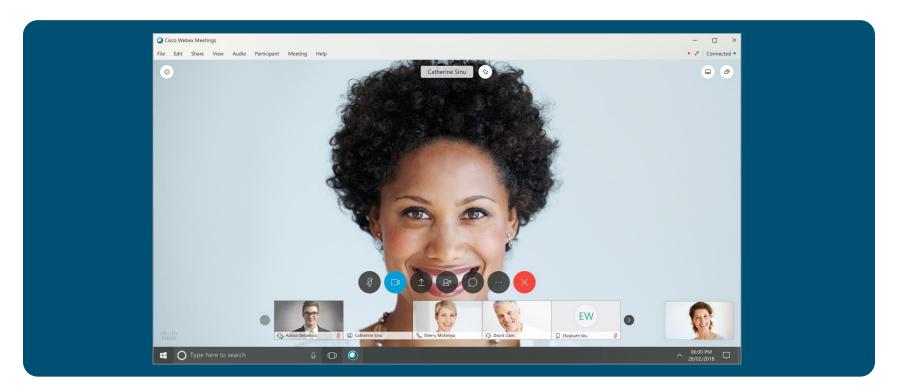


Autonomy to customize video and content layout



In-meeting control bar

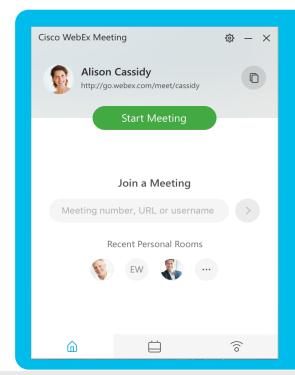
Intuitive and one-click for most common tools

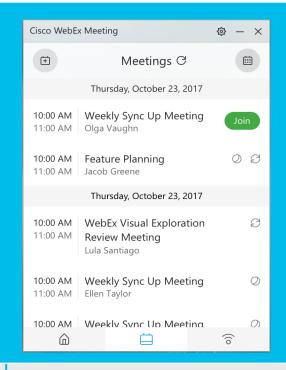


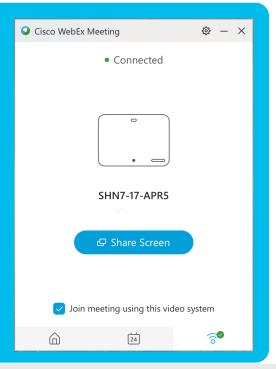


Webex Teams Modular App

Fastest Join Experience





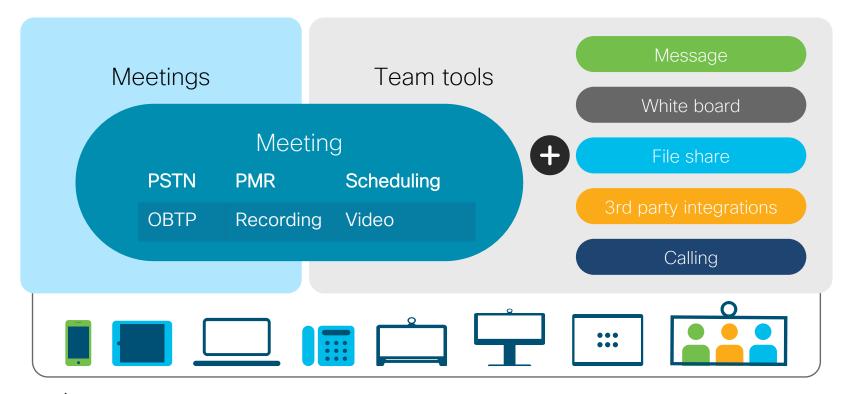


One click to start or join

Integrated calendar

Proximity for device pairing and content share

Optimized for the way You work





A New Meeting Experience Mobile and Web Optimized





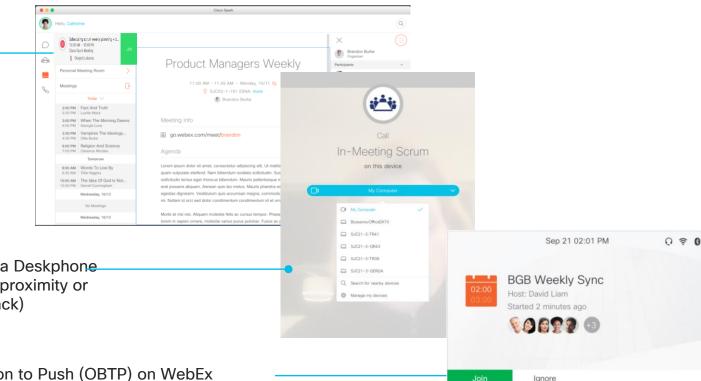
Reimagined mobile meetings equal participation

Using WebRTC – no plug-ins to host or join a meeting



Joining Meetings Made Easy

- One Button to Push (OBTP) to join WebEx, Teams or other SIPbased meetings
- Requires Cisco WebEx Teams Calendar services (Exchange, 0365, Google calendar)



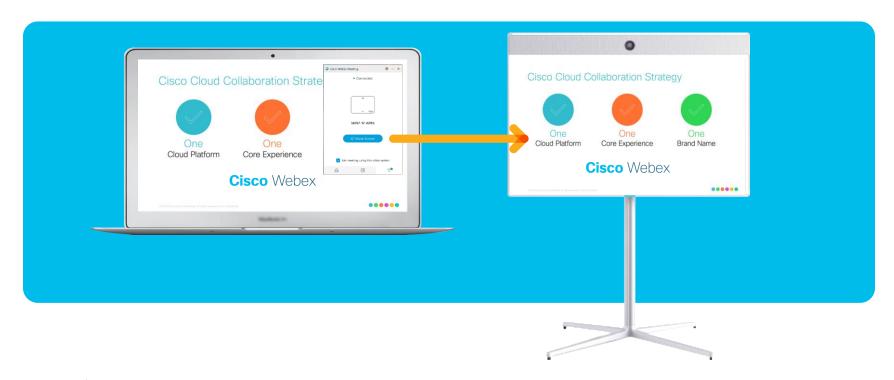
 Device selection to join via Deskphone control, WebEx device in proximity or PSTN / TP (dial in / callback)

> One Button to Push (OBTP) on WebEx Shared and Personal Devices

cisco / ise/

Intuitive and impactful meetings

Detect and connect to your video device for wireless sharing and control with Cisco Proximity



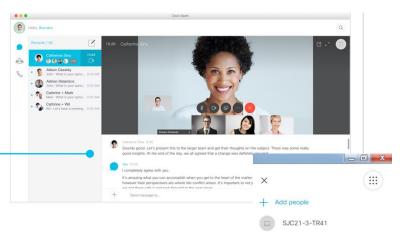
Bringing Context To Your Meetings

- See who is in the meeting so you can decide when to join!
- Easily switch from a meeting to another



 Multitask or locate the file you need to share in the meeting

> Identify who is sitting in the conference room (WebEx devices)

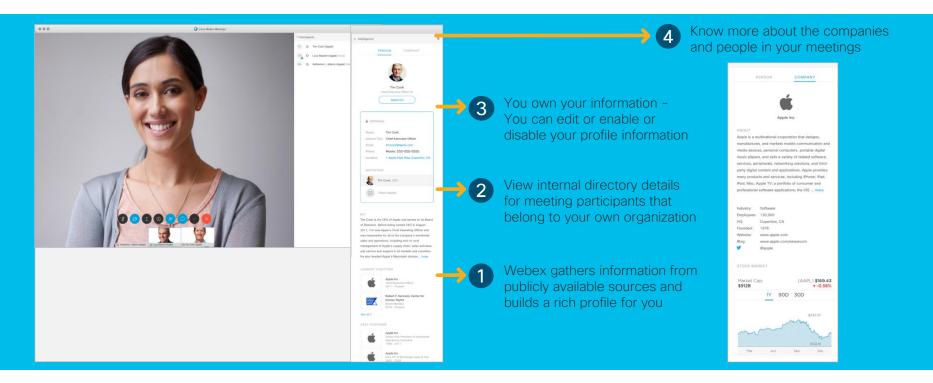




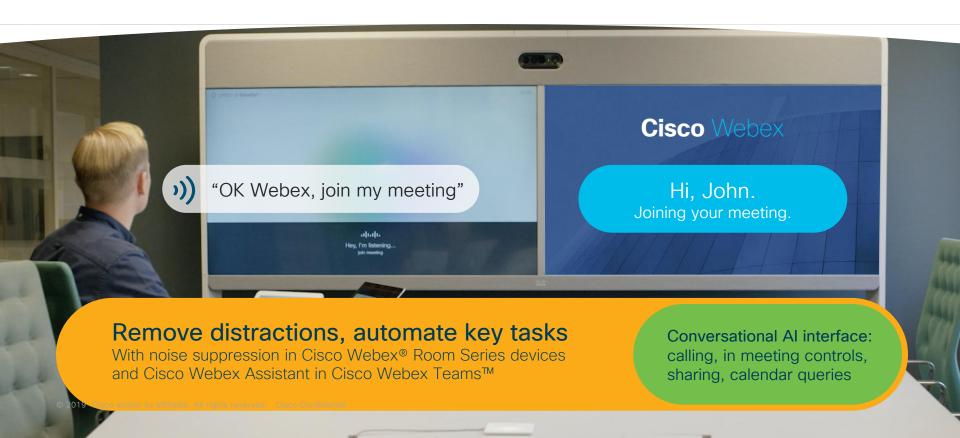
Gail Batac

Cognitive Collaboration: People Insights for Webex

Learn more about the people in your meeting



Cognitive Collaboration: Webex Assistant and Noise Suppression Al powers smarter ways to work



Cognitive Collaboration: Facial Recognition

Know who you're talking to



Meet where you work

From your calendar

From messaging apps

From where you learn

From your car

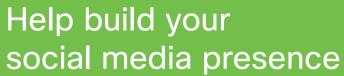


Enable the best meeting experience from third-party tools



Stream to your followers

Facebook Live integration



Stream your video feed directly through Facebook Live





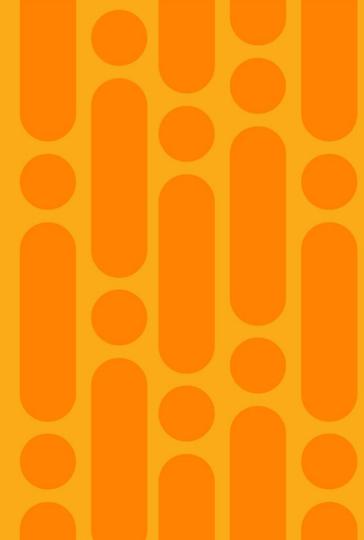
Additional Cisco Webex Services

Webex Events	Highly engaging, large-scale ¹ online events to global audiences
Webex Training	Highly interactive and effective online training and e-learning ²
Webex Support	Real-time, efficient, personalized customer service and IT support
Webex Webcasting	Managed service for large-scale online events (up to 40,000+ participants)

Cisco Webex® Assist Services available for meetings, events, and trainings

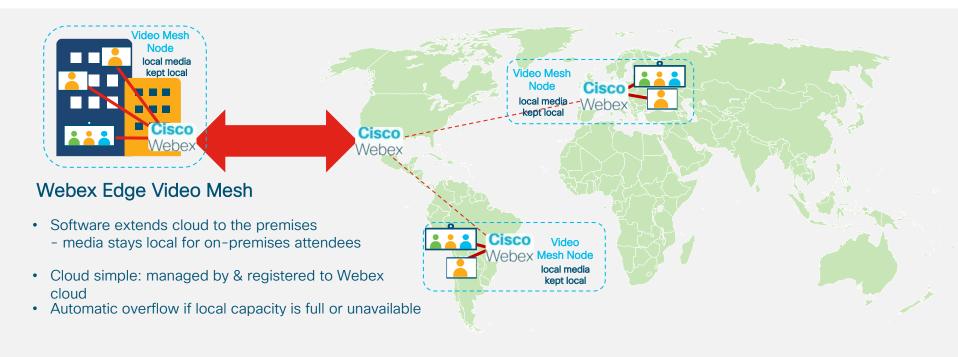


Video Mesh



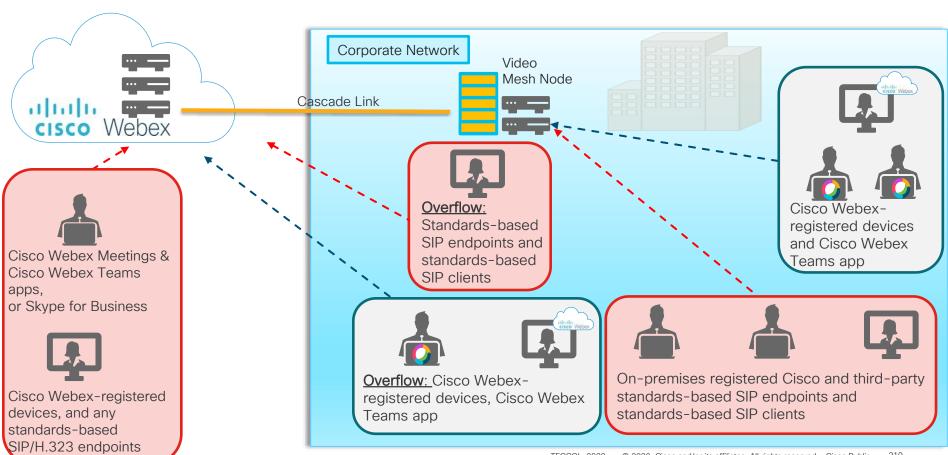
Webex Edge Video Mesh

On-premises video quality and bandwidth savings





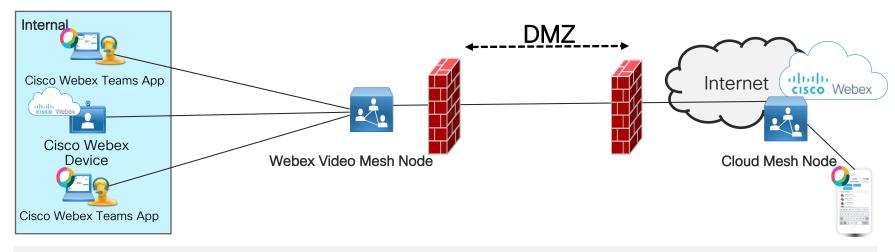
Cisco Webex Video Mesh



Architecture

Webex Video Mesh - Option 1





Internal installation considerations:

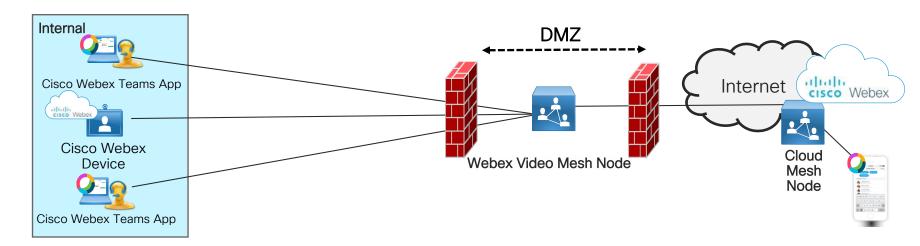
- All media for internal participants stay internal
- Placed with other collaboration infrastructure devices
- Single connection per conference to Cloud Mesh Nodes



Architecture

V

Webex Video Mesh - Option 2



DMZ installation considerations:

- External media does not traverse the internal network.
- All media for internal participants goes to the DMZ.
- Security policy does not allow Cisco Webex network ports to be opened outbound for media directly to the Internet from the internal network.



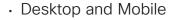
What devices and scenario can the Video Mesh node be used?

Uses the Node



- Any Cisco Webex-registered device
 - · SX, MX, DX, Room-series, Webex Board









- CUCM-registered devices
 - Calling a Cisco Webex scheduled restring c personal room.
 - SX, MX, DX, Room-series, Jabber, Jabber VDI (12.6 or higher)

Uses the Node

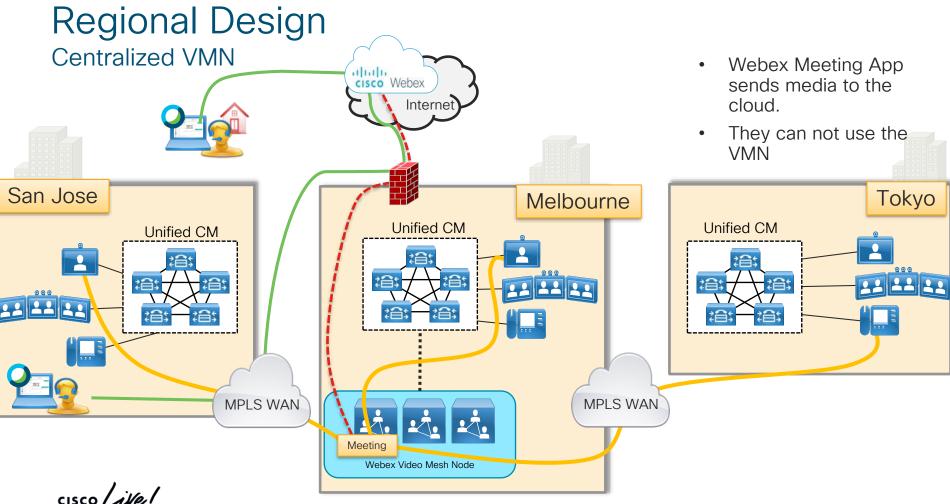


- · Cisco VCS/Exp.-registered devices
 - Calling a Cisco Webex scheduled meeting or personal room.
 - · SIP or H.323 (requires Interworking)
- Cisco Webex VDI client (39.3 or higher)
- Cisco Webex Call My Video System to Webex-registered endpoints









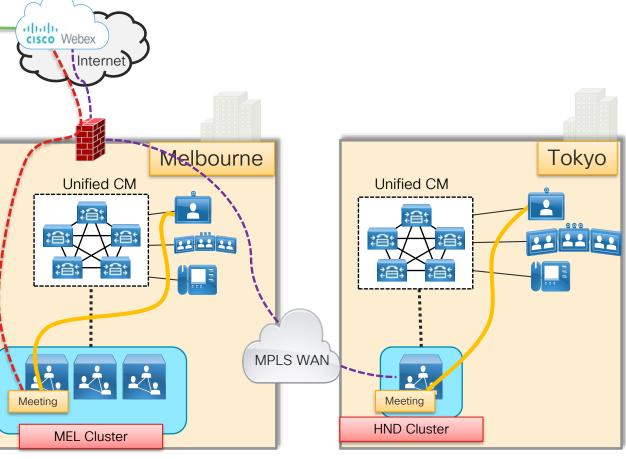
Regional Design

Two Clusters

Video Mesh is a hub and spoke architecture

Hub is the Webex cloud

Spokes are the clusters





Number of ongoing calls

Node Type, SW Image Usage and Service Status

Node specific Alarm **Notification**

Cisco Webex Video Mesh Node











Overview





calls

Node Details

Video Mesh Node Type Production Image

Deployment Type

Release Channel Beta

Provisioning Version

2019.07.12.1917m.1

Cms1000

Cloud

OS Version 2135.5.0 OoS On

Maintenance Mode Off

-\- Node Health

CPU 70 cores, 0.33% used

Memory

3 66GB of 58 96GB used (6.2%)

13,43GB of 251,56GB used Disk Space (6%)

Management

Service

NTP Sync

Active

Messaging Service

Active

Notifications

△ Hostname configuration of the Video Mesh Node is invalid.

The Video Mesh node's hostname settings are invalid and these alerts were generated on the node: Unable to resolve FODN videomesh1.ucdemolab.com's IP address against the current DNS settings. These issues

may cause sharing problems for users in Webex meetings. Please check the hostname

and domain settings Last Reported: Wed Jul 24 2019 13:08:33 GMT-0400

(Eastern Daylight Time)

△ Experienced problem connecting to Cisco

Experienced problem connecting to Cisco

Network Information

용 Network Settings

videomesh1.ucdemolab.com Hostname

00:50:56:88:bb:73 Interface 10 99 255 21/24 10.99.255.1 Gateway

DNS 10.99.101.12.10.99.102.11

ntp.ucdemolab.com, 10.99.101.12,

10.99.102.11 Dual IP

NTP

Disabled

Registration Details

Registered

Org ID

Cluster

Cluster ID

Yes UC Demo Lab Organization

35a15b0a-0ef1-4029-9f63-

a7c54df5df59

Boxborough - Full

b7c30185-99cc-4ba7-b5b5-

77adb1871dd8

Cloud Connectivity

Webex Cloud Resolution Test 3rd Party Resolution Test

Webex Cloud Connectivity Test

3rd Party Connectivity Test

Webex Cloud Bandwidth Test 3rd Party Bandwidth Test

Pass Pass

Pass

Pass Testina

Webex Cloud services.

Last Reported: Tue Jul 16 2019 00:44:22 GMT-0400

(Eastern Daylight Time)

Webex Cloud services

Node Registration Status

DNS Test, Server Response time, BW test

Hover to reveal when the test was run and what was checked

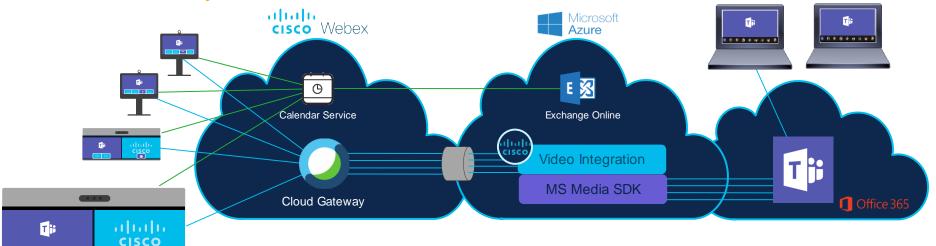
Microsoft Teams and Cisco Webex



Cisco Webex Video Integration for Microsoft Teams

Cloud Gateway

ű:



Cisco & Third- Party SIP Cloud and Premises Registered Schedule & Join: Calendar integration provides OBTP on Cisco video endpoints. One click to join from Microsoft desktop and mobile

In-meeting: flexible multi-screen layouts, MS Teams users see their respective layouts, receive and send content, plus meeting controls

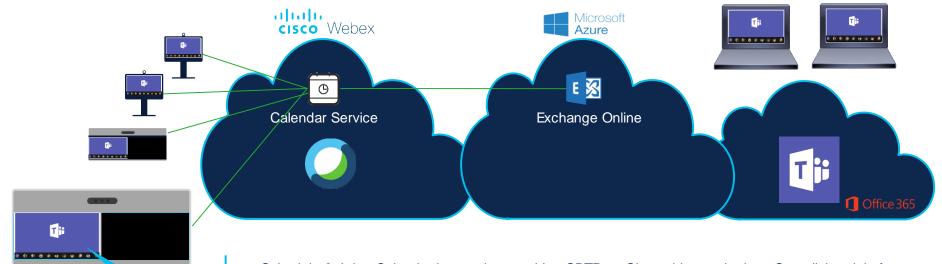
Participant List: Cisco and Microsoft participants receive full participant roster

Set Up: Administered via Webex Control Hub and Office 365 Admin



Cisco Webex Devices for Microsoft Teams meeting

Microsoft Web App (WebRTC)



Schedule & Join: Calendar integration provides OBTP on Cisco video endpoints. One click to join from Microsoft desktop and mobile.

In-meeting: Microsoft Web App single screen layouts, receive content share, no content share via PC/HDMI cable (future), controls limited (end call, local mute, volume)

Deployment: Cisco Video endpoints must be cloud-registered or registered on-premises using Webex Edge for Devices

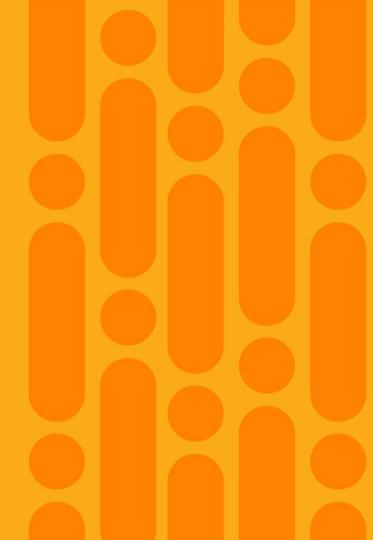
Set Up: Administered via Webex Control Hub and Office365 Admin



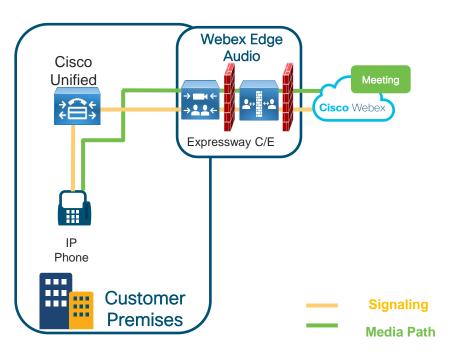
Cisco Webex Device

(Latest Generation)

Webex Edge



Architecture requirements



Unified CM support only

10.5 or later

Cisco UCM registered IP phones

• Supporting G.711 or G.722

Expressway support only

- X8.10 or later
- Can use existing Expressway C/E deployment
- Audio scale dependent on Expressway deployment and services enabled.

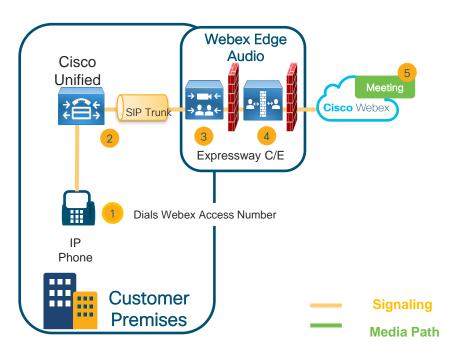
Webex site

- WBS 33.x or higher
- Included in Flex, A-WBX and A-SPK SKU need the Webex Edge Audio package
- Not available on CCA-SP, CCA-ENT or TSP sites.
 - Requires migration to Webex Audio Site

Requires a signed certification from a Cisco trusted Certificate Authority (CA)



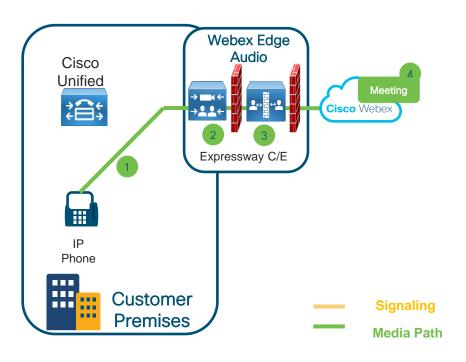
Dial in Signaling Call Flow



- Endpoint dials Webex Audio access number.
- 2. Cisco UCM matches the number and routes as +E.164 through SIP trunk to Expressway-C.
- 3. LUA script on SIP trunk to Expressway-C applies transformations required for correct routing to Webex
- 4. Expressway-C sends request to Expressway-E.
- 5. Expressway-E routes call to the Webex cloud.
- 6. Meeting resources are setup.



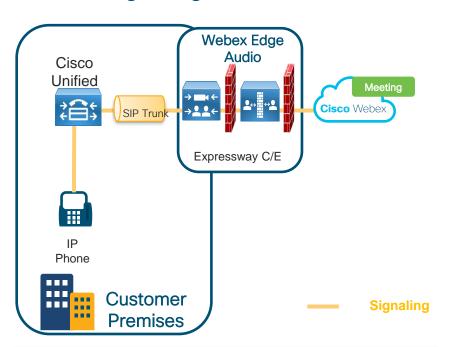
Dial in Media Call Flow



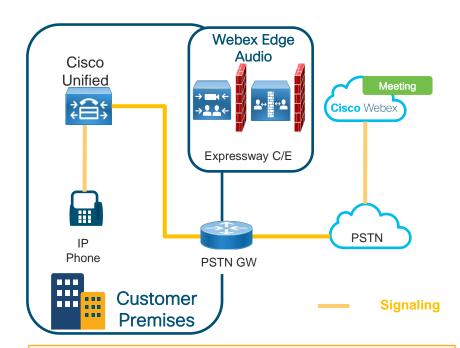
- The IP phone sends media to Expressway-C
- 2. The Expressway-C sends media to Expressway-E via the traversal zone
- 3. The Expressway-E sends media to the Webex cloud.
- 4. IP phone's audio is mixed into the meeting and it hears the other participants.



Dial in Signaling Call Flow with and without Edge Audio configuration



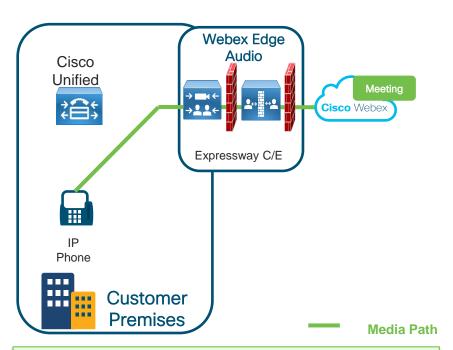
User dials E.164 Webex Access number that **does** have an associated Edge Audio route pattern in Cisco UCM



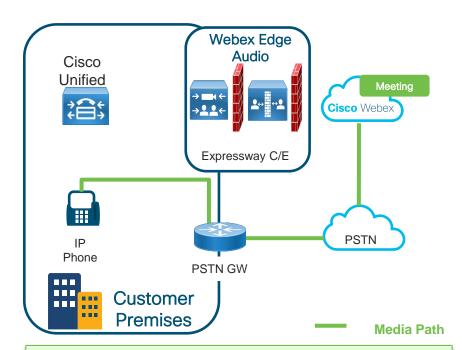
User dials E.164 Webex Access number that **does not** have an associated Edge Audio route pattern in Cisco UCM



Dial in Media Call Flow with and without Edge Audio configuration



User dials E.164 Webex Access number that **does** have an associated Edge Audio route pattern in Cisco UCM



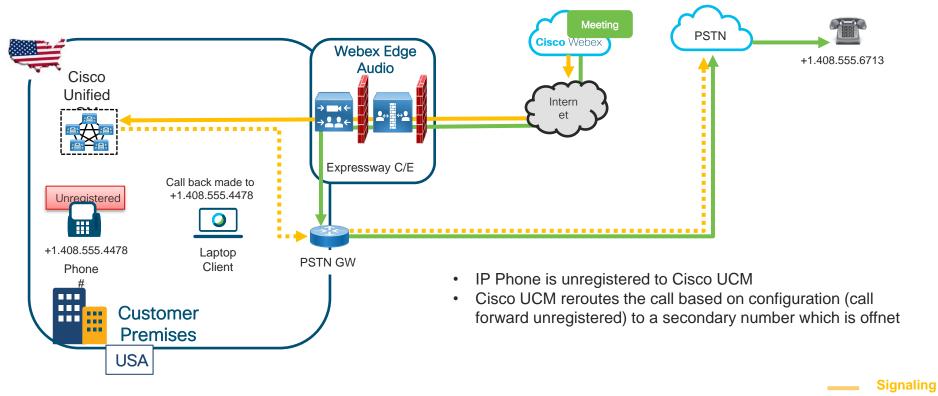
User dials E.164 Webex Access number that **does not** have an associated Edge Audio route pattern in Cisco UCM



Deployment Scenarios



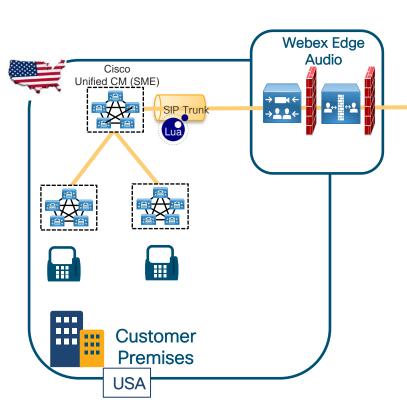
Single site Call Back



Media Path

TECCOL-2982

Single site with SME

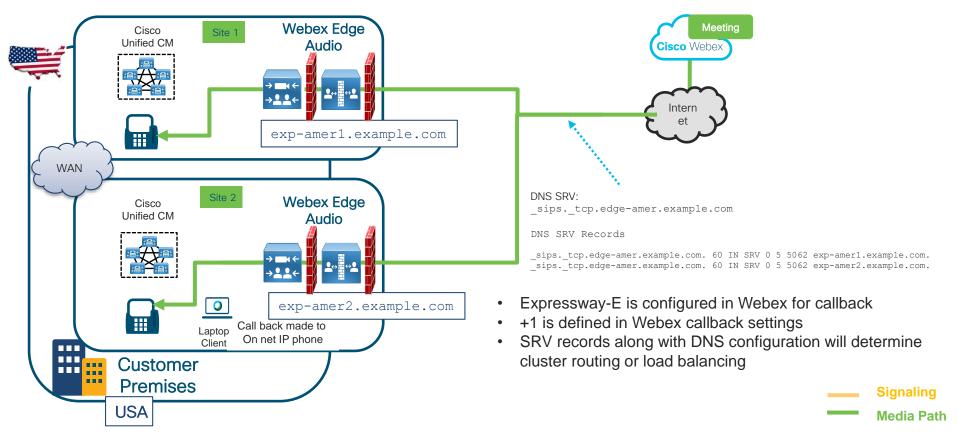


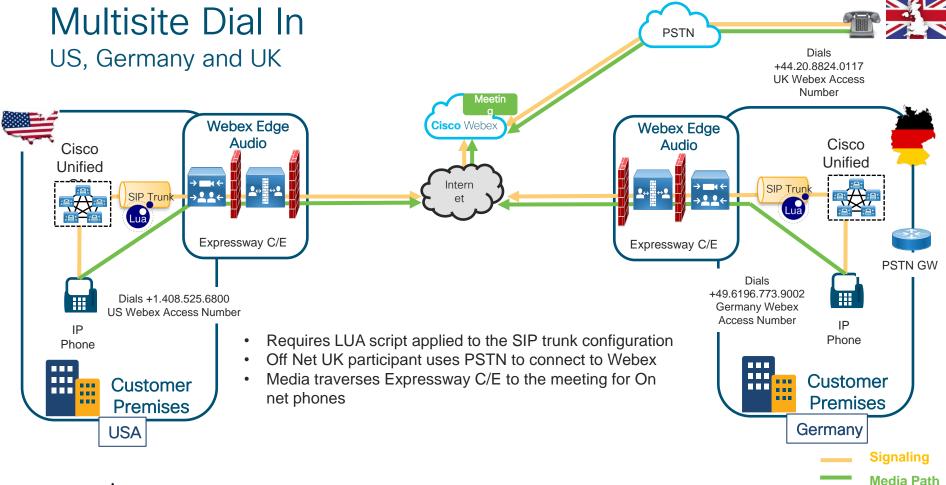


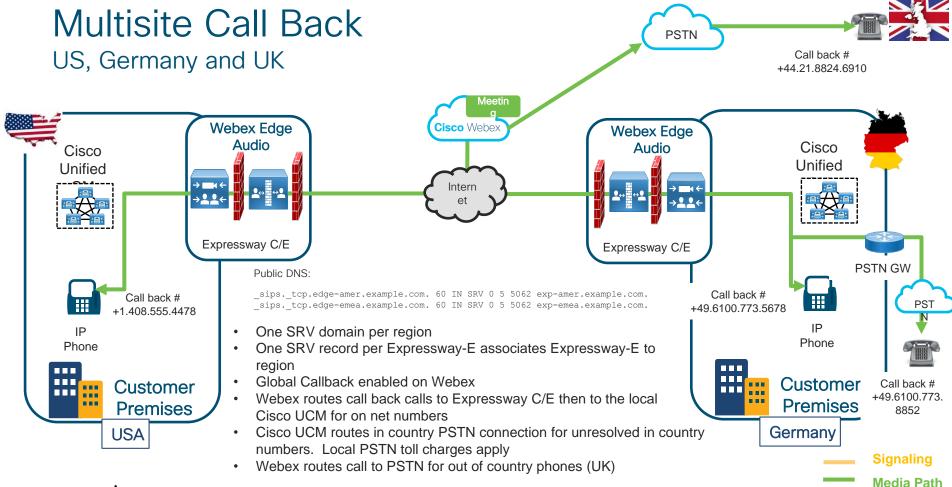
- Session Manager Edition is supported
- +E.164 enterprise dial plan to route dial-in and call-back
- Inter-cluster routing using ILS/GDPR
- Apply LUA script on SME trunk for dial-in
 - Request and To: URI manipulation
- Enterprise dial plan can support arbitrary dialing habits for dial-in
 - ... as long as the number ultimately exposed to LUA script is a valid +E.164 Webex dial-in number



Single Country Call Back - Multiple Expressways

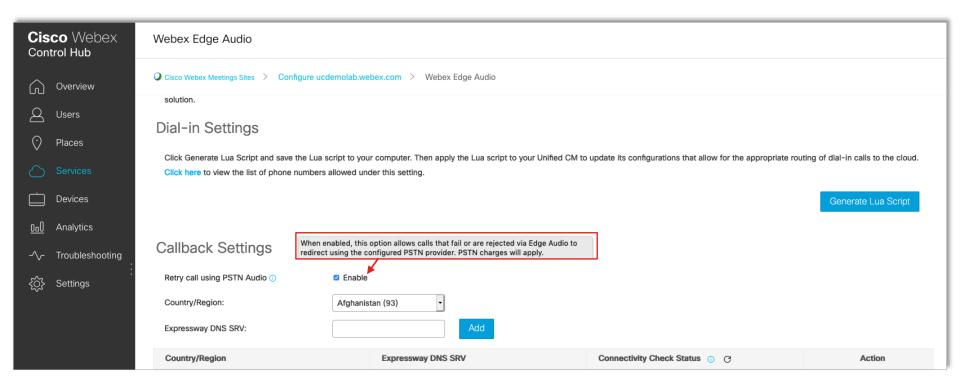






PSTN Fall Back





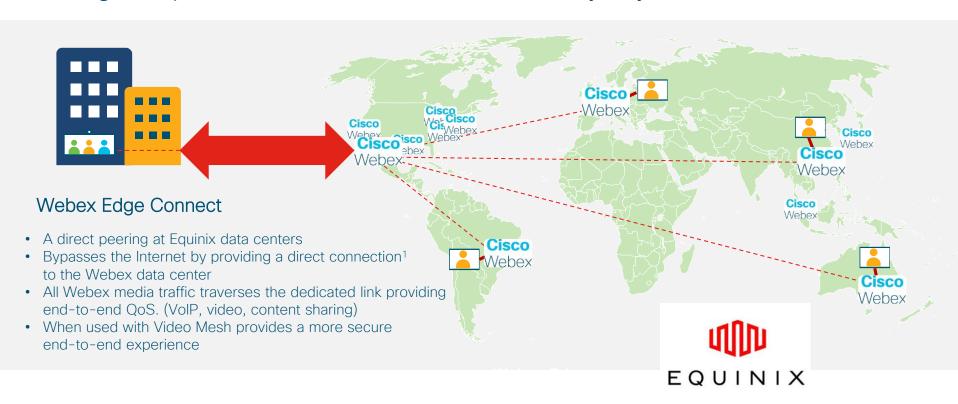


Webex Edge Connect



Webex Edge Connect

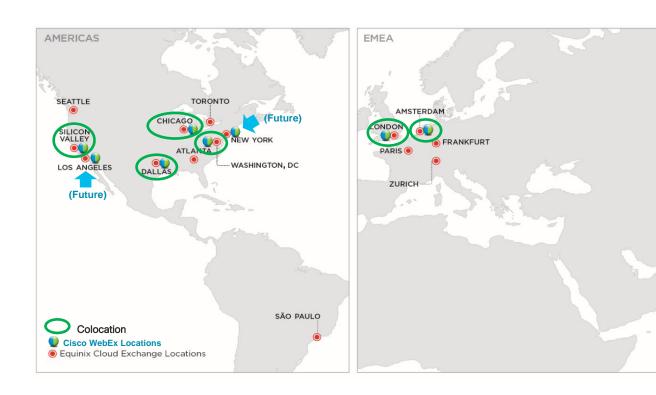
Brings the power of the Webex backbone directly to your data center



cisco Life!

¹ via a peering agreement with Equinix

Webex Edge Connect and ECX Locations

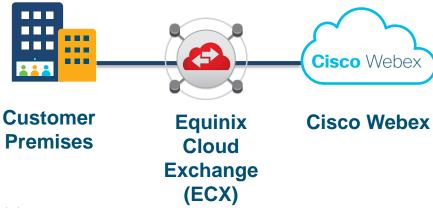






Customer Requirements

- 1. A cage and router in place at Equinix
- 2. A paid connection to the Equinix Cloud Exchange
- 3. Knowledge of BGP Routing
- 4. Public BGP Autonomous System Number
- 5. Public provider independent IP block
 - No RFC1918 addressing (10.0.0.0/8, 172.16.0.0/12, 192.168.0.0/16)
 - Customer may rent a /29 IP block from Equinix
- 6. Paid service to Cisco Webex



Architecture

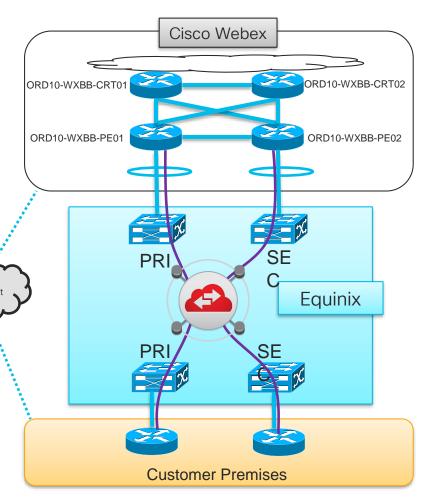
 A customer setups up dual connections to Equinix for redundancy

 Cisco Webex has redundant connection to Equinix at all colocations across the globe

 BGP routing is used to route traffic across the peering connection.

 Customers that have a global presence can choose which regions to peer.

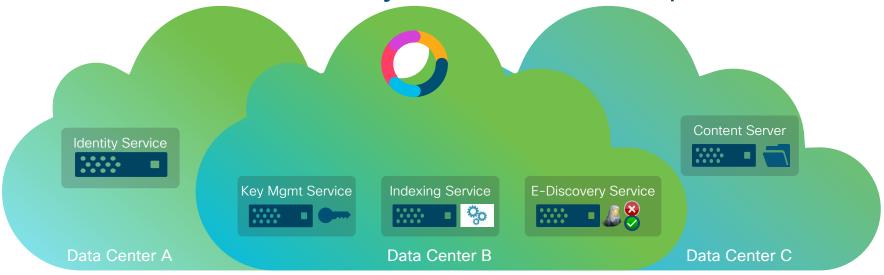
Customer's Internet connection is used as fallback



Cloud Security and Hybrid Data Security



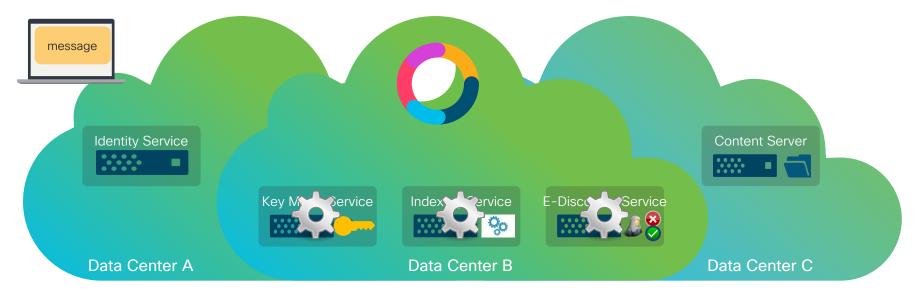
Webex Cloud Security - Realms of Separation



Webex logically and physically separates functional components within the cloud Identity Services holding real user Identity (e.g. email addresses) are separated from Encryption, Indexing and E-Discovery Services, which are in turn separated from Data Storage Services



Realms of Separation - Encryption and Storage



Webex logically and physically separates functional components within the cloud Data Services such as Encryption Key Generation, Secure Message Indexing for Data Search, and E-Discovery functions operate in different Data Centers from the Data Center that encrypted content is stored in

cisco Live!

Realms of Separation – Identity Obfuscation



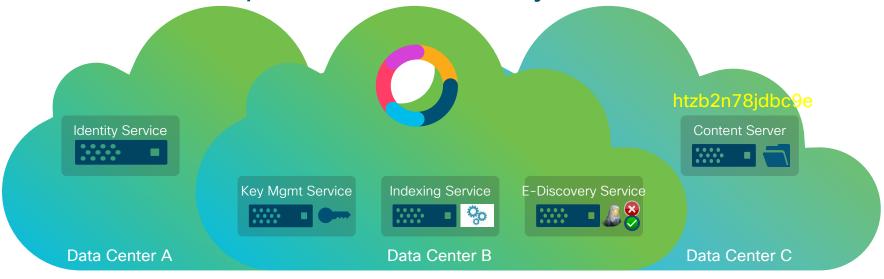
Outside of the Identity Service - Real Identity information is obfuscated:

For each User ID, Webex generates a random 128-bit Universally Unique Identifier (UUID) = The User's obfuscated identity

No real identity information transits the cloud



Realms of Separation - Identity Obfuscation



Outside of the Identity Service - Real Identity information is obfuscated :

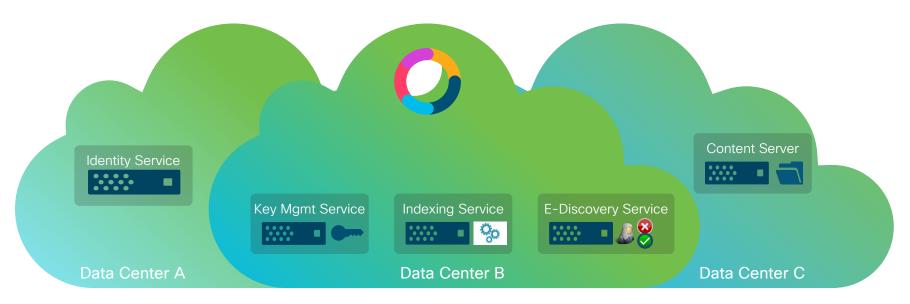
For each User ID, Webex generates a random 128-bit Universally Unique Identifier (UUID) = The User's obfuscated identity

No real identity information transits the cloud

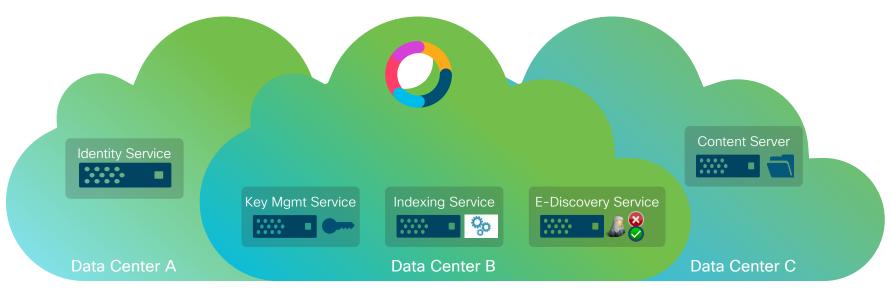


Customer Controlled Security: Hybrid Data Security





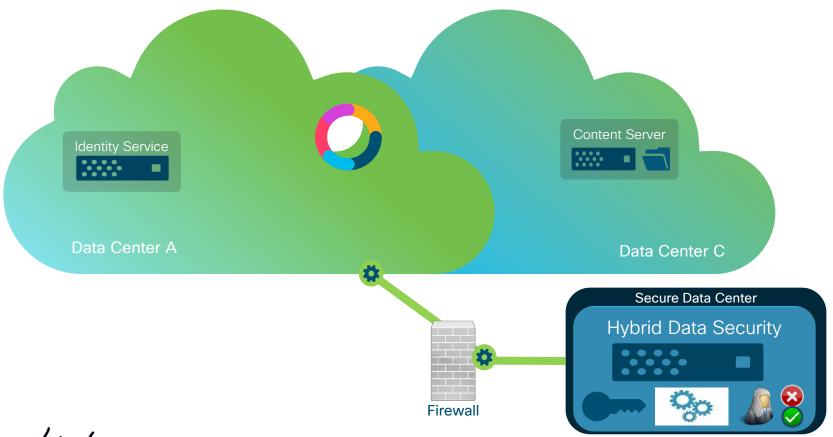


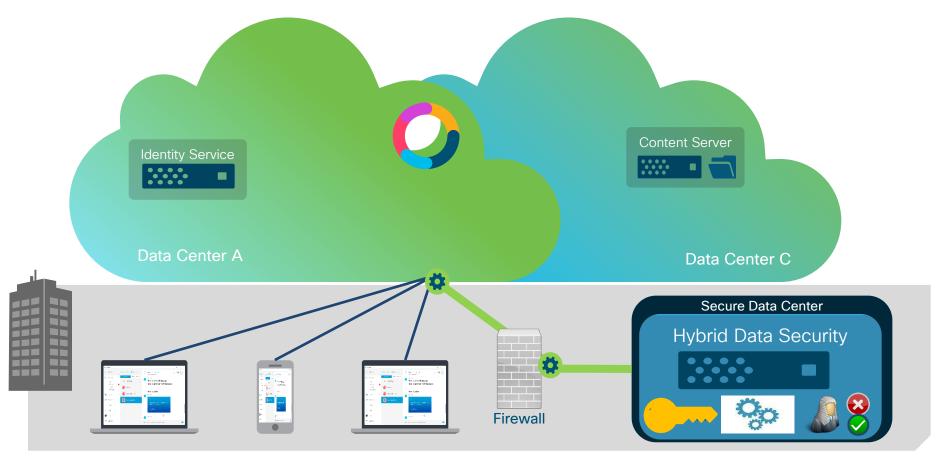


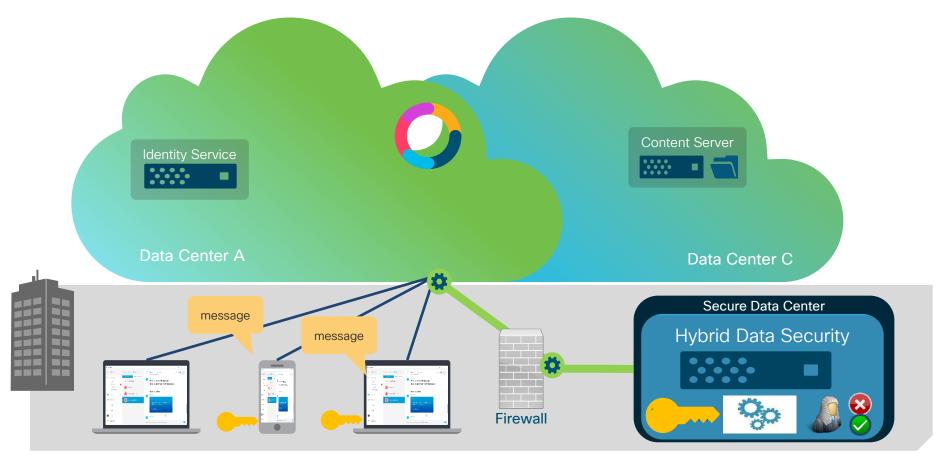




TECCOL-2982



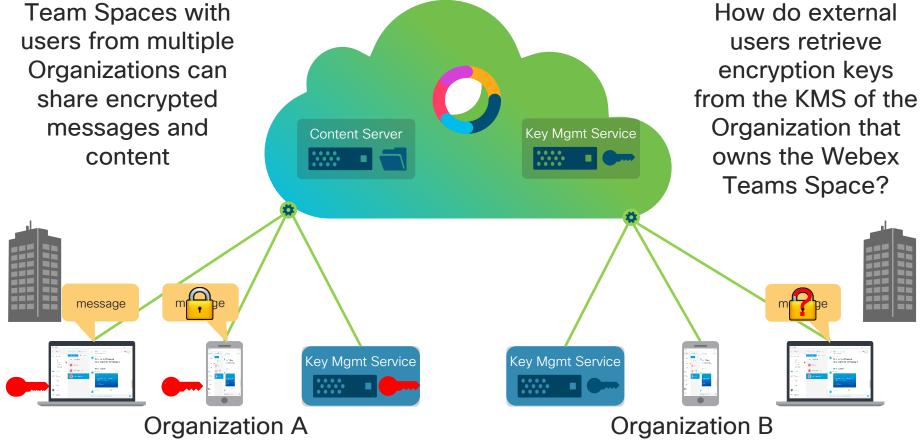




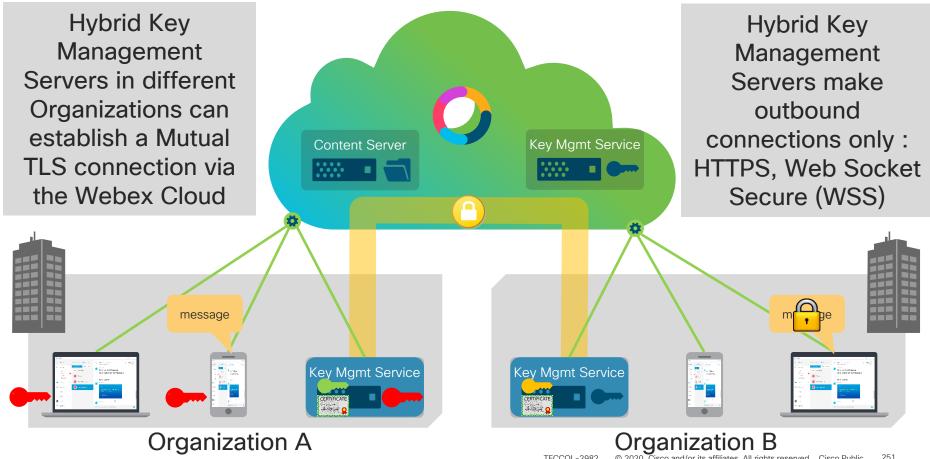
Customer Controlled Security: Key Management Server Federation



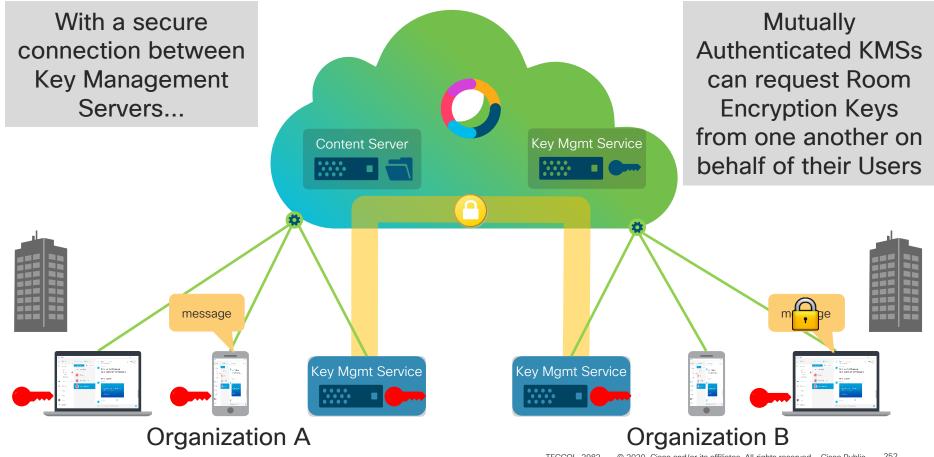
HDS: Encryption Keys & Users in other Orgs



HDS: Key Management Server Federation



HDS: Key Management Server Federation

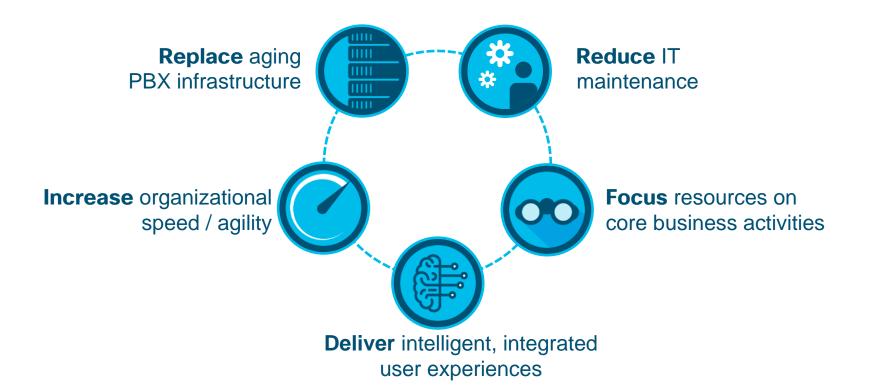






cisco live!

5 reasons to move your calling to the cloud



cisco Live!

Choose the enterprise solutions that meet your needs



Cisco Webex Calling



Cisco UCM Cloud

Midsize businesses and large enterprises:

Cisco operated multi-tenant cloud

Proven enterprise cloud PBX

Ultimate simplicity

OTT / Internet

Large enterprises with sophisticated migration requirements:

Cisco hosted, private instance

UCM/Jabber features

Custom IT integrations

Private Network / Equinix



Sophistication —









Key elements

- Packed with the features and applications your business needs
- Simplified all-in-one package
- Part of Webex® the world's **favorite** collaboration suite 54% market share*

*Synergy Research, Dec. 2018

Benefits of Webex Calling

Secure,
enterprise-grade
cloud calling from
a trusted brand

Offers a more intelligent collaboration experience

Enables a more responsive, agile organization

Management without the complexity

Makes IT strategic again

Complete, integrated UCaaS functionality

Webex Calling

- Fully featured PBX functionality
- Seamless Webex Teams and Webex Meetings Integration
- Apps, open SIP, and Cisco integrated device support
- Webex Teams centered experience
- Full PSTN connectivity
- Unique, native mobile integration
- Secure, redundant, carrier grade global cloud

Webex Teams

- Secure group messaging
- Persistent team spaces
- Native file sharing with option to use Microsoft OneDrive/SharePoint
- White boarding in the app and on the Webex Board
- Rich integrations: Microsoft, Google, Salesforce, Asana, Jira, Trello, ServiceNow...
- Video meetings on any device with interop to SIP and Skype for Business users

Webex Meetings

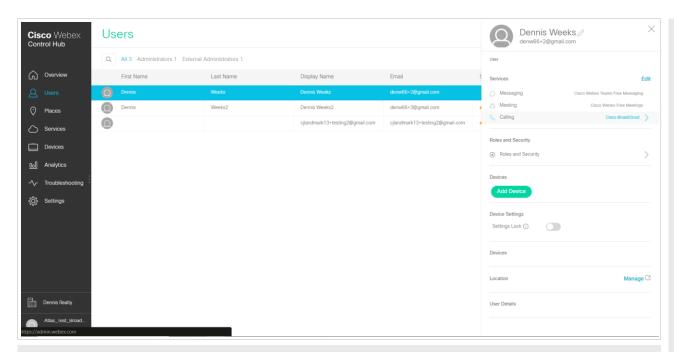
- World's leading meetings service
- Award winning Webex video devices
- Secure, redundant, managed, QoS optimized global network
- Leading mobile experience: customizable video views, native screen sharing, hands-free joining
- Intelligence: Webex Assistant, People Insights, one-push/pairing join
- Rich integrations: Google, MSFT, Slack, LMS, Ford cars, etc.

Common Identity, Calendar, UX, Management, Licensing, Ordering



Alternate numbers	Analog hotline	Anonymous call rejection	Auto attendan	t Barge-in	exempt	Business continuity (CFNR)
Busy lamp monitoring	Call forwarding always	Call forwarding busy	Call forwarding no answer	Call forwarding selective		Call history
Call forwarding no answer	Call hold and resume	Call logs w/click to dial	Call notify	Call re	edial	Call routing and queuing
Call return						Call waiting ID
Conferencing (site based)	All the enterpris features your		seamless ation	Addresses your RFP requirements		nterprise phone directory
Extension o	business needs	to the	cloud	RFP require	rements	bound caller ID
Main number outbound call II						-Way calling (6)
Office anywhere Feature-rich business telephony						Priority alert
Push-to-talk	Remote office	Schedules	Sequential ring	Reception	ist client	Reports and metrics
Selective call acceptance	Selective call rejection	Shared line	Simultane	ous ring	Speed Dial 100	
T-38 Fax support	Three-way calling	Unified messaging	Video (point to po	oint) Visual vo	icemail	Voice mail

Easy administration with Webex Control Hub



Users can be provisioned, entitled, and configured by either the partner or customer administrators

Users are managed from the user panel in Webex®

- Entitlement, number assignment, roles, devices, and location can all be configured
- Call-specific configurations can be edited by the end user in self-care or by the administrator in the Webex Calling admin portal

Three Webex Calling "station types"

Enterprise

- Full-featured business solution
 - Full set of PBX telephony features
 - Webex Teams™
 - Webex® Meetings (optional)
- Associated with a person



Basic

- For phone-centric office users
 - Full set of PBX calling features
 - · Limited mobility features
 - Only "free" version of Webex Teams
- Appropriate for cubicles and shared desk locations
- Associated with a person or place

Common area

- Simple dial tone for common area phones
 - Limited telephony feature set
 - No Webex Teams or Meetings
- Appropriate for locations like lobby, breakroom, etc.
- Associated with a place





Global availability



Delivered globally



- Three regional platformsNA, EMEA, and APAC
- Offer services to multinational customers from a single region
- Secure data and traffic within chosen region

Data center committed expansion:

- Geo-redundant DC in Japan (2Q-CY2019)
- Data center in Canada (1Q-CY2019)

Data center considered expansion:

- Second data center in EU (Amsterdam, CY2019)
- Second data center in U.K. (CY2019)



Market availability

Contract and satellite countries





Contract countries

Australia **Finland** Luxembourg Austria France Malta Mexico* Belgium Germany Bulgaria Greece Netherlands Canada Hungary New Zealand Ireland Croatia Norway Cyprus Italy Romania Czech Republic Japan* Poland Denmark Latvia* Portugal Estonia* Lithuania* Slovakia

Slovenia*
Spain
Sweden
Switzerland
U.K.

Satellite countries

Algeria* Macau* Angola* Malaysia Mexico Argentina Azerbaijan* Singapore Benin* South Africa* Bosnia & South Korea* Herzegovina* Taiwan* Brazil Thailand* Hong Kong* Turkey* Indonesia Vietnam* Kenya*

Relationship between Cisco® and partner only in contract countries

U.S.

^{*} Target availability as contract, or satellite country, as indicated, in August, 2019



Contract versus satellite countries

Selling versus deploying



Contract countries

"The country where the partner sells to enterprises. The partner is on-boarded in the contract country, covering items such as interconnects, billing, invoicing, language, and currency. The partner's service is active in the regional platform for their contract country."



Satellite countries

"The country where the partner deploys enterprise offices. It is the partner's obligation to ensure it meets any regulatory requirements. Satellite countries have varying degrees of localization."

Relationship between Cisco® and partner only in contract countries

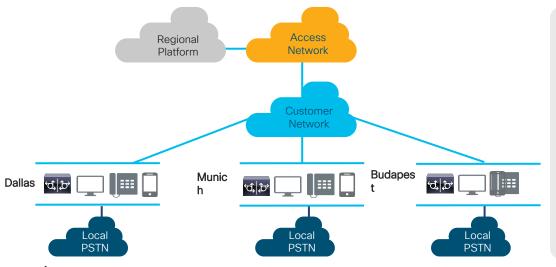


Example: Webex Calling contracted in the U.S.

Acme is a U.S. Webex Calling partner with an enterprise customer, Digital Design, that is headquartered in Dallas with branch offices in Munich and Budapest.

Acme signs a U.S. Webex® Calling agreement. Acme provisions Digital Design's Dallas, Munich, and Budapest offices in CCW. Full local language and localization support exist for the U.S. and Germany. Local dial plan and device localization only are supported for Hungary. Hungarian users can use any supported Webex Calling language.

Cisco's relationship with Acme is in the U.S. only (contract country) - not in Germany or Hungary (satellite countries).



Service configuration

Regional platform

North America

Contract country

- **United States**
- USD

Satellite countries

- Germany
- Hungary

Data center locations

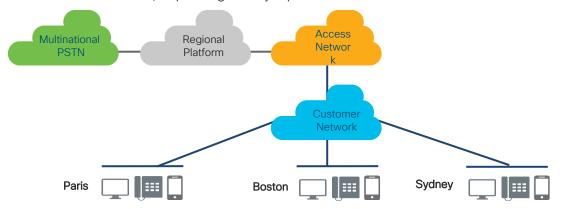
U.S.

Example: Webex Calling contracted in France

Acme's enterprise customer, EU Brokers, is based in Paris with offices in Boston and Sydney. EU Brokers requires its data to stay inside the European Union.

Acme signs a French Cisco Webex® Calling agreement. Acme provisions EU Brokers' Paris, Boston and Sydney sites, with all data staying in the EU.

Note that this is a subsequent Webex agreement to Acme's existing U.S. one. This enables Acme to operate a U.S.-based and French-based service, depending on buyer preference or market need.



Service configuration

Regional platform

FMFA

Contract country

- France
- EU

Satellite countries

- United States
- Australia

Data center locations

Europe



Clients and devices



Cisco voice device portfolio

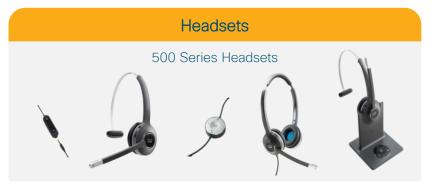
Collaborate your way with a device for any user, in any location





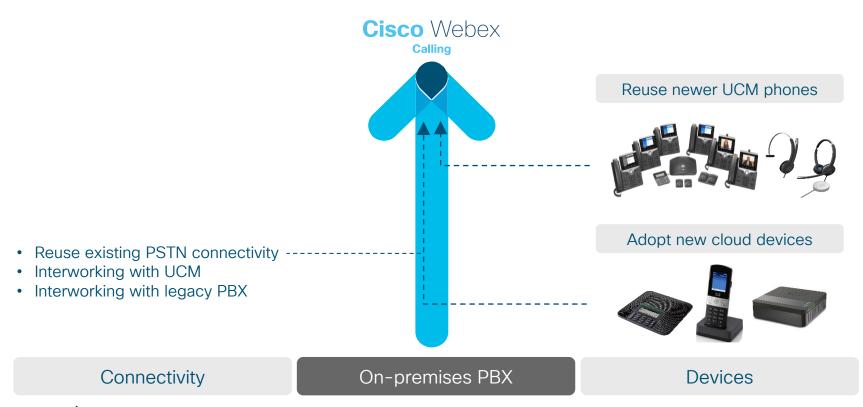
TECCOL-2982



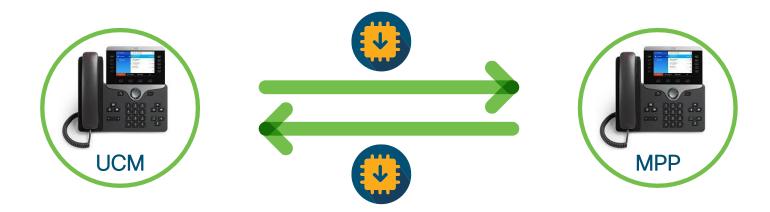




Protect investments in Cisco IP phones



Phone firmware migration





Migration firmware

- All 7800s: Some hardware limitation (pre-CY2016)
- All 8800s except 8821, 8831, 8851NR, 8865NR
- KEMs do not require migration
- Partial data loss call history, local contacts



Migration license

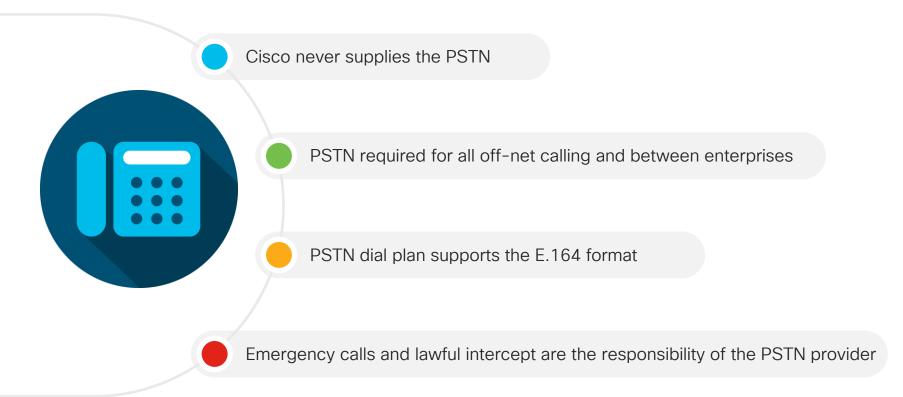
- Flex plan includes 1 license per user
- Locked to MAC address
- One-way migration per license



PSTN options



Webex Calling: PSTN quick facts



Webex Calling PSTN options







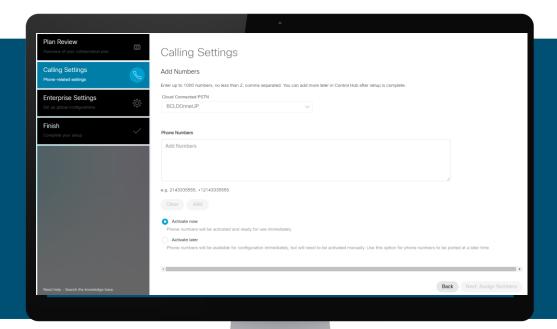
BYO Carrier ••••••



Multi-PSTN connectivity

Increased flexibility, PSTN coverage and cost optimization

- Support multiple PSTN providers per partner
- Define PSTN providers at the location level





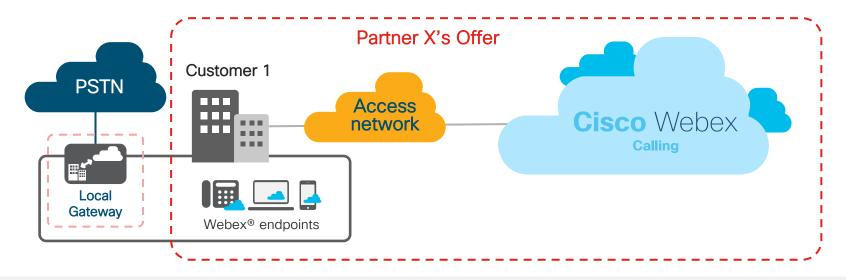
Webex Calling PSTN: Cloud Connected



- PSTN access through peering with a selection of Cloud Connected partners integrated into the Cisco Webex®
 Calling cloud
- Partner helps enterprise procure and provision PSTN
- · Available to service providers who serve customers with locations outside of their PSTN footprint



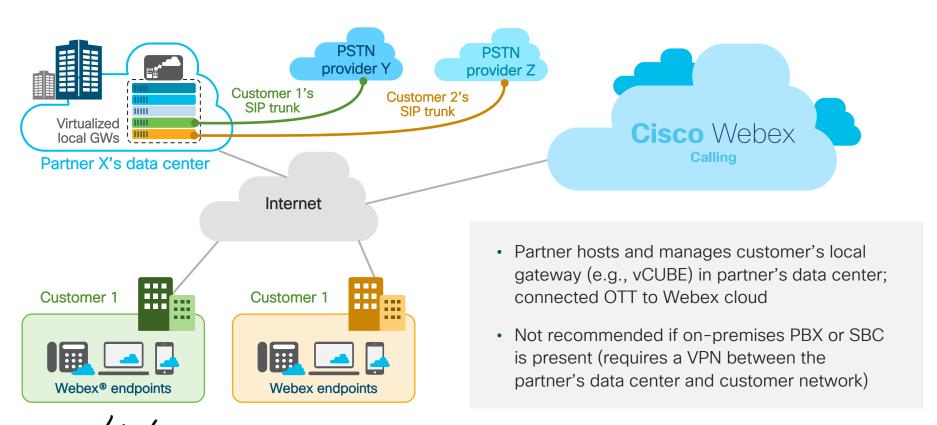
Webex Calling PSTN option: Local Gateway PSTN through customer premises



- PSTN access through a Local Gateway device at the customer site and the customer's PSTN service (SIP Trunk, PRI, etc.)
- PSTN service decoupled from Cisco® Webex Calling service

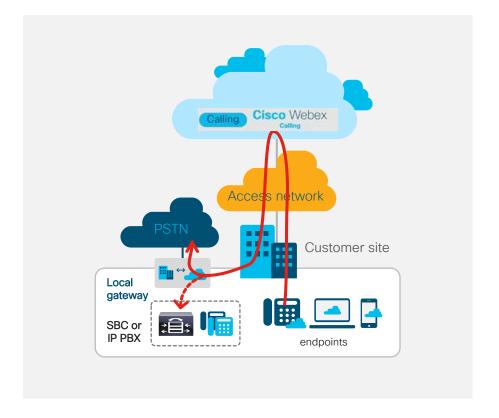


Webex Calling PSTN option: Local Gateway PSTN through customer premises



Local gateway Architecture overview PSTN option

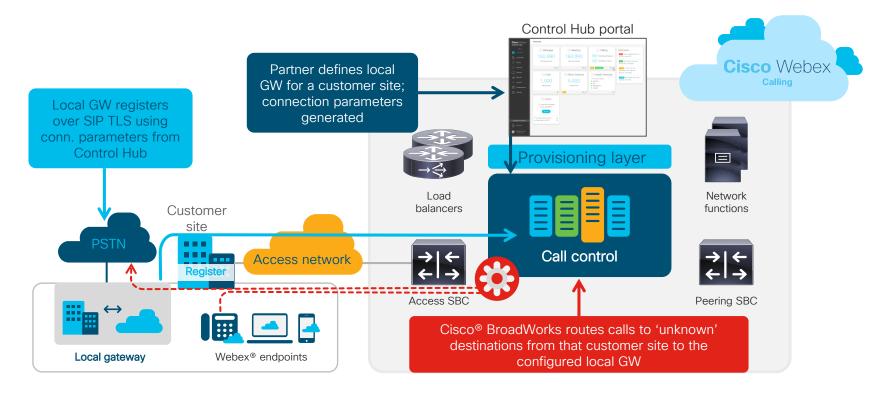
- Enables a BYOPSTN option for Webex Calling
- Provides connectivity to a customer-owned PSTN service
- May also provide connectivity to an on-premises IP PBX or SBC





CUBE as local gateway

Architecture overview





Call routing with local gateway



Enterprise dial plan enhancements



Familiar dialing habits for enterprise users



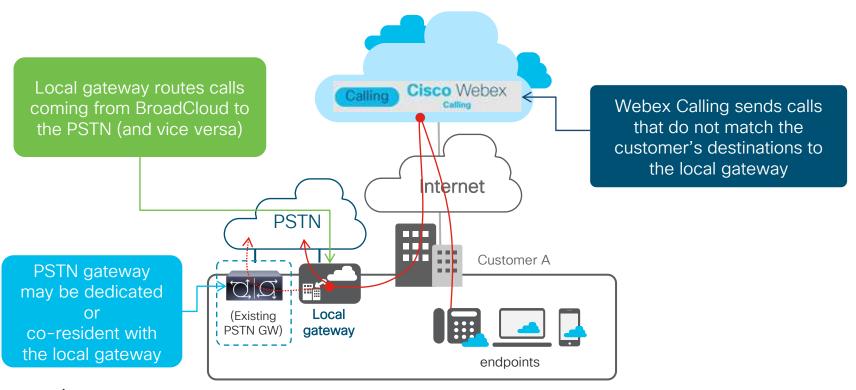
Basic enterprise dialing use cases to support Webex® Calling



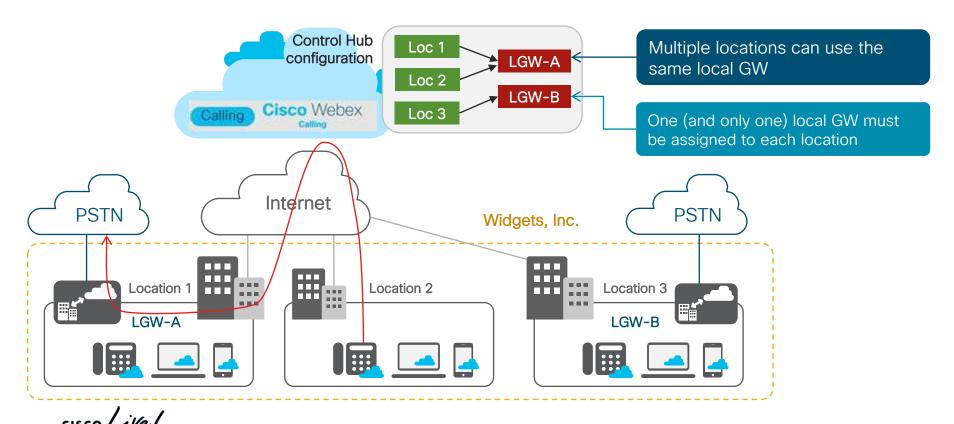
- Ability to have a steering digit for PSTN calls (9 to dial out)
- Permissive dialing allowed from the Webex Calling app (with or without an access code)
- Ability to define a prefix length for site and extension ranges
- Ability for off-hook dialing without using the SEND key or waiting for interdigit timeouts



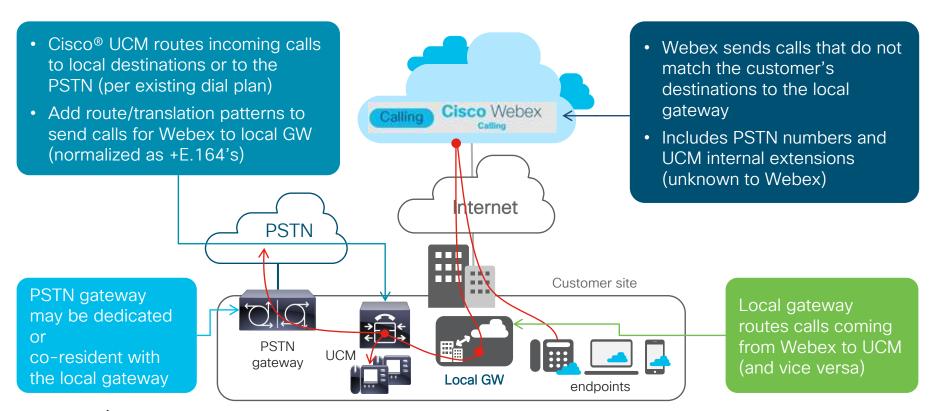
Call routing Single local gateway



Call routing Multiple local gateways

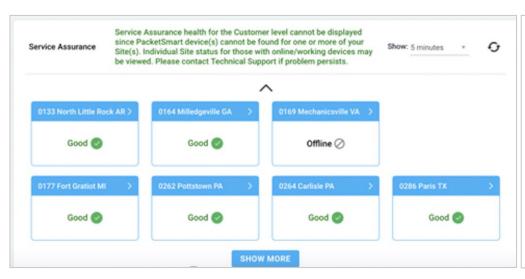


Call routing With an IP PBX/CUCM



Service assurance for Webex Calling

- Overall view of site-by-site media quality for Webex Calling sites accessible by partner and enterprise admin
- Browser-based pre-qualification bandwidth and port testing tool







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Webex Edge for Calling



Solution: Connect, Centralize & Unify



CONNECT

Begin the migration to the cloud with Webex Calling for branches & knowledge workers with site to site calling across all calling platforms

CENTRALIZE

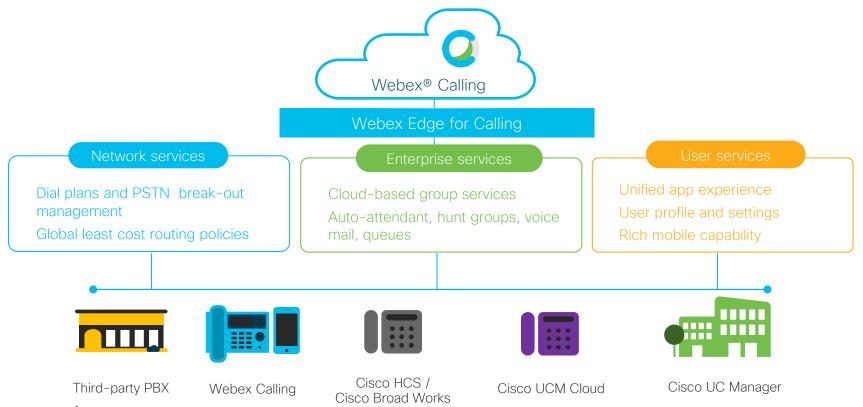
Centralize calling routing with a cloud-based dial plan across Cisco & 3rd party PBXs to simplify management and lower costs while leveraging existing investments

UNIFY

Unify and optimize the experience through a single client to deliver a set of consistent collaboration services across your enterprise

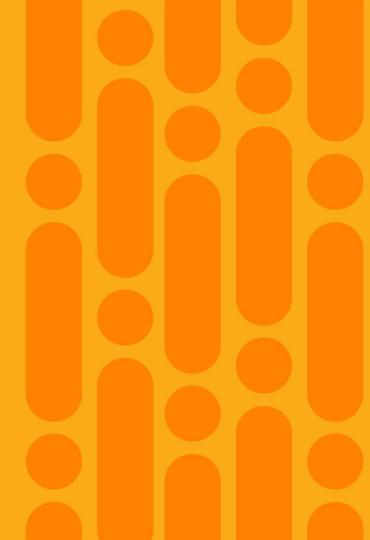


Webex Edge for Calling: Key capabilities



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UCM Cloud



Cisco UCM Cloud

Private cloud services delivered over a dedicated infrastructure to a single enterprise



Cisco Unified Communications Manager

- Enterprise unified communications and collaboration as a service from the Webex® Cloud
- Dedicated resources and security controls provide a stronger cybersecurity posture
- Scalability, agility, and efficiency of a resilient private architecture assures your business-critical services will always be available and operational
- Enhanced levels of controls and customization capabilities
- · Ability to integrate into existing IT systems to ensure business continuity
- Powered by the Cisco® Unified Communications Manager in the cloud



Cisco Unity®
Connection



Cisco Jabber®



Cisco Webex Teams™



Cisco Webex® Meetings

Cisco UCM Cloud





Hosted and operated by Cisco; sold and managed by partners

Enterprise-tested Unified Communications Manager as a service from the Cisco® Cloud

- Powered by the proven Cisco® Communications Manager platform with over 100 million devices worldwide offered as a service from the Webex® Cloud
- Hosted and operated by Cisco; sold and managed by partners
- An enterprise-ready calling solution with all the features, integrations, and benefits of Cisco UC Manager plus enhanced security options capable of meeting FedRAMP authorization
- Designed for enterprises and public sector customers
- Webex Teams[™], Webex[®] Meetings, Jabber[®], and all collaboration endpoints are supported
- Cisco UCM Cloud is available in the Cisco Collaboration Flex Plan as a cloud calling option



Cisco UCM Cloud positioning

1K+

Cisco UCM Cloud

- Generally intended for large enterprise customers (1000+)
- Minimum initial order can be as small as 1000 to support ramping
- Possible exceptions for smaller customers are not a fit for Webex® Calling

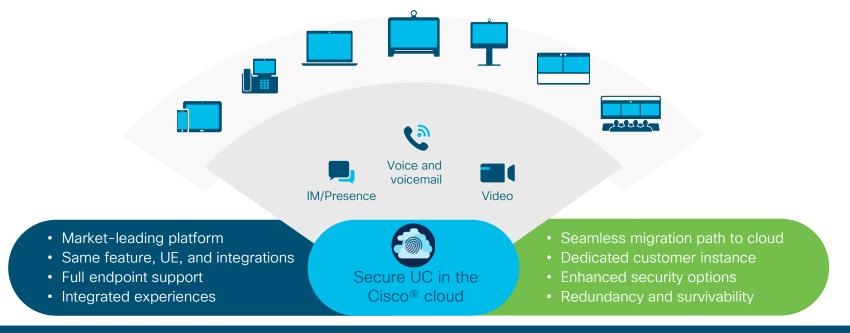
Cisco UCM Cloud

- UCM Feature Set
- Commercial and Enterprise
- Customizable Private Instance
- Global Deployments
- FNCRYPTFD!

Webex Teams™ and Webex® Meetings are available with both services



Cisco UCM Cloud calling key elements Currently based on Cisco 12.5 Collaboration Solution Release (CSR)



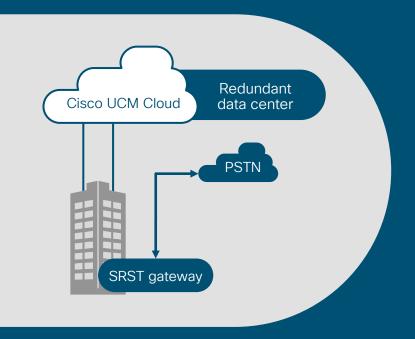
All the benefits of a traditional phone system with integrated collab tools without the management overhead

Strategic features for Cisco UCM Cloud transitions

Cisco UCM compatibility benefits	
100% Cisco® UCM-compatible features	Cisco endpoint support (new and legacy voice and video devices; Cisco UCM version-dependent)
Support for complex dial plans and media encryption	Cisco Contact Center interoperability (Webex® Contact Center, UCCE, UCCX, HCS-CC)
Native Cisco TelePresence® and video (registration, B2B video, and PSTN dialing)	Highly customizable
Private cloud (SD-WAN, MPLS, etc.) with on-net media optimizations	LDAP, Active Directory, SSO, MFA
Cisco voice gateway support (ASR, ISR, VG)	Cisco survivability support (MGCP, ASR, ISR)
Virtualized Desktop Infrastructure (VDI)	Advanced emergency call handling (Cisco Emergency Responder)
IT integrations (CRM and extensive third-party ecosystem)	Computer Telephony Integration (CTI)
Bulk provisioning and extensive APIs	Native peering for private connection
Proven scale for large customer deployments	PSTN flexibility (partner-provided or bring your own)

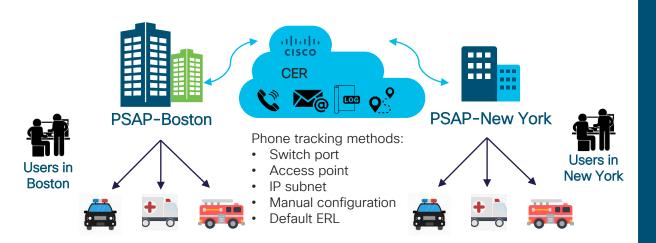


Cisco UCM Cloud survivable telephony



- Cisco® UCM Cloud provides proven redundancy via two Webex® data centers and dual links per data center
- Survivable Remote Site Telephony (SRST) adds automatic telephony survivability during outages for seamless transitions during failure and restore
- Available for both headquarters and branch locations

Cisco Emergency Responder



- Routes emergency calls to a local Public-Safety Answering Point (PSAP)
- Alerts staff via email or phone of a local emergency call for quick response
- Logs all emergency calls
- Provides PSAP with accurate dispatch address for the caller in need
- Automatic location tracking for soft clients and phones

PSTN options



Partner Bundled

 Partner interconnects to UCM Cloud and is responsible for bundling UCM Cloud Calling and PSTN minutes



Local Gateway

 Customer or partner-managed edge device for PSTN interworking and legacy PBX interworking (including Cisco® UCM)



Bring Your Own SIP Trunk

Partner helps customer procure and provision PSTN,
 via certified SIP trunk provider interworking directly to UCM Cloud



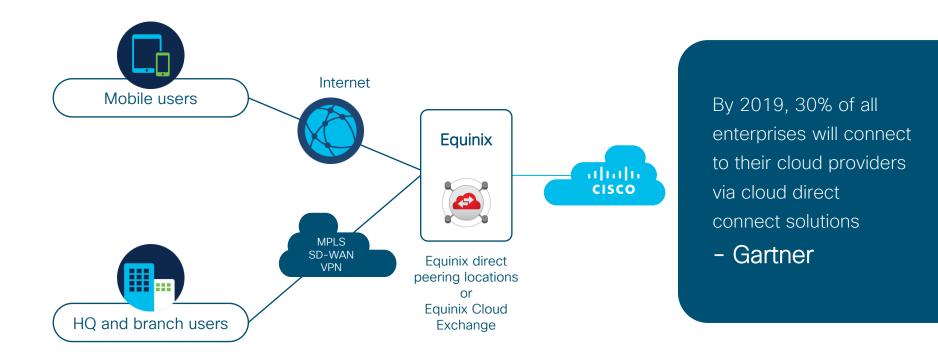
Preferred Media Partner (PMP)*

 Partner helps customer procure and provision PSTN, via pre-interconnected media partner to UCM Cloud

*Roadmap



Cisco UCM Cloud connectivity



Cisco UCM Cloud - global data center availability



Cisco® Unified Communications Manager Cloud (Cisco UCM Cloud) Cloud-optimized, point-to-point media traversal architecture reduces the need for multiple regional data centers

Service availability

Worldwide

Service is not available in U.S.-embargoed countries:

- Crimea
- Cuba
- Iran
- North Korea
- Syria

The UCM Cloud partner is responsible for the regulatory compliance of its customers

- Due to constantly changing global telecommunications laws and regulations, service availability may become restricted in some
 countries. While Cisco routinely monitors applicable telecommunications laws and regulations in an attempt to adapt to changing
 legal environments, we reserve the right to modify our country coverage listing as necessary to meet all country legal
 requirements.
- Brazil, India, Russia, and Singapore have unique regulatory requirements and require evaluation on a case-by-case basis by Cisco[®] legal representatives.



Simplified device administration

Centralized device management



- Centrally manage Cisco[®] endpoints, headsets, video devices, and Jabber[®]
- User-based configuration
- Inventory summary and reports
- · Centralized telemetry and analytics

Simple and secure phone activation



- Activation code issued via email
- Phone authenticates using the activation code
- UCM Cloud config update is automatic and a phone registers securely
- Simple user experience. Reduced admin. Secure. Lower onboarding cost. Bulk capabilities.



Phone activation – unified end-user experience

1. Generate activation code



- Admin generates for users
- Available on the self-care portal

2. Unbox and plug in



3. Enter activation code

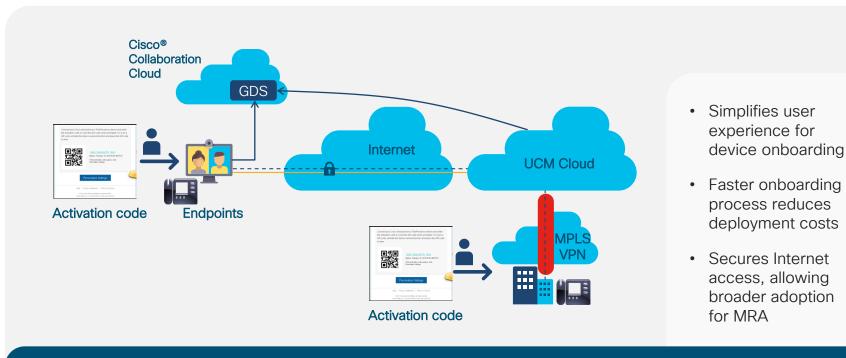


- QR code for video endpoints
- Via numeric keypad for other phones

4. Ready to go (secure)



Activation code-based device onboarding



7800 and 8800 Series IP phones - 7811, 7821, 7832, 7841, 7861, 8811, 8841, 8845, 8851, 8851NR, 8861, 8865, 8865NR



Fully centralized device management

Headset management

Centrally manage Cisco® Headsets from Unified Communications Manager:

- User profile templates
- Managed firmware releases
- Customized inventory queries

No additional end-user software **No** new infrastructure

No additional licensing



No more XML file uploads!

Create Jabber® configuration templates – per site, per user group, etc.

Jabber client can retrieve configuration settings from the service profile



Full provisioning support for video endpoints

Intuitive tabbed admin interface

Simplify - migrate video endpoints from VCS to UCM

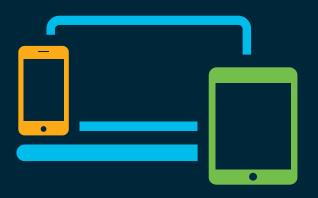
Easily restore the configuration in the event of a factory reset or RMA/ spare swaps

*Video management feature supports CE 9.4 and above: Cisco MX Series, Cisco SX Series, Cisco Webex® DX Series, Cisco Webex Room Series





Complete your online session survey

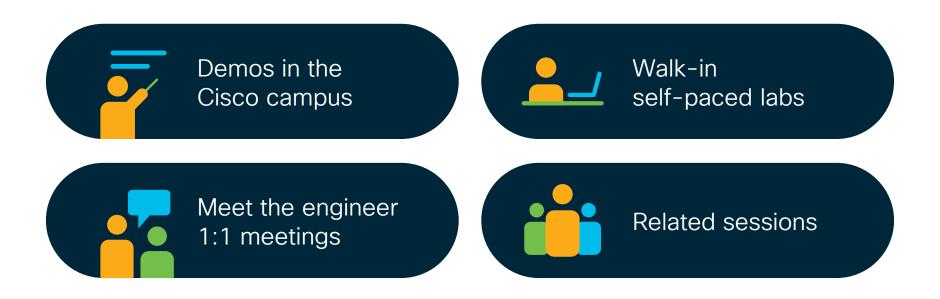


- Please complete your session survey after each session. Your feedback is very important.
- Complete a minimum of 4 session surveys and the Overall Conference survey (starting on Thursday) to receive your Cisco Live t-shirt.
- All surveys can be taken in the Cisco Events Mobile App or by logging in to the Content Catalog on <u>ciscolive.com/emea</u>.

Cisco Live sessions will be available for viewing on demand after the event at ciscolive.com.



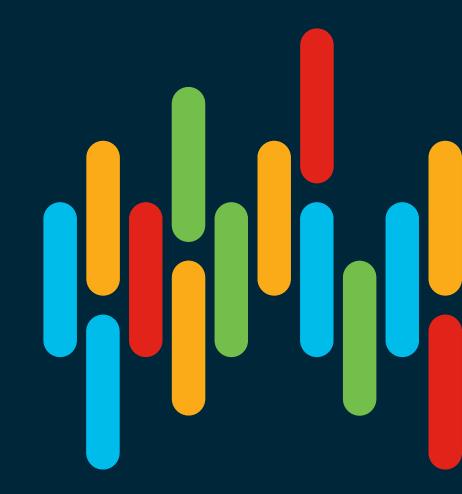
Continue your education





illiilli CISCO

Thank you



cisco live!





You make possible