Test Results for Cisco Unified Communications System Release 10.0 for Japan

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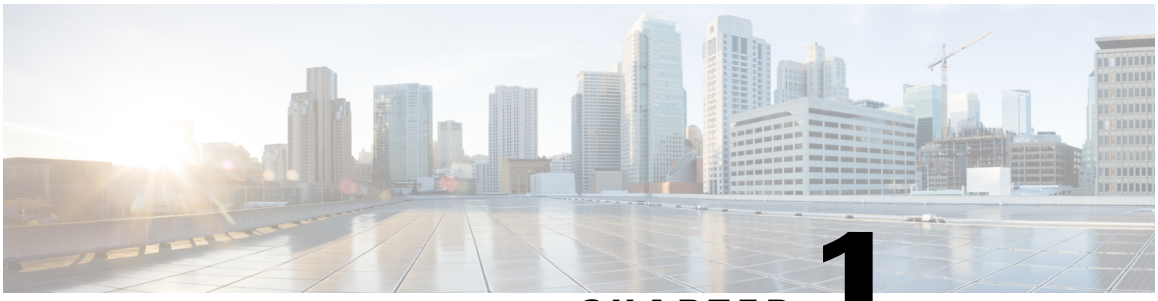


CHAPTER 1

Cisco Unified Communications System Test

Cisco Unified Communications System Test

Cisco Unified Communications System Test, an integral part of the Enterprise Voice Solution Management is a program that validates and tests specified systems-level solution for the various products and platforms in the Cisco Unified Communications System.

Cisco Unified Communications System Test, the systems integration layer, ensures that the Unified Communications components delivered across the various engineering teams when combined, improves the Unified Communications System software quality. This is achieved by testing the various components.

The requirements for Cisco Unified Communications System Test is derived based on the following:

- Popular customer scenarios
- Input from various Business Units, fields and Cisco Services

The test bed architecture is built based on the Solution Reference Network Design (SRND), cross-section of product deployment models etc. The different types of testing carried out as part of Cisco Unified Communications System Test are:

- Interoperability/Compatibility
- Functionality
- Availability/Reliability/Stability
- Performance/Scalability/Capacity
- Usability, Serviceability
- Special focus area - CAP (Customer Assurance Program), Technical Assistance Center (TAC)
- Security

Cisco Unified Communications System Test for Japan

Cisco Unified Communications System Test for Japan, in turn is an add-on testing at the solution level, where the requirements gathered are specific to Japanese usage and market. The requirements are derived based on the following:

- Customer found defects in selected UC products
- High priority cases that are covered by the Cisco Unified Communications System Test team
- Inputs from SEs, TAC team of Cisco Japan

The test execution is carried out on selected UC products, which affect the Japanese segment and that are prioritized by SEs of the Cisco Japan team. Japanese specific equivalents, such as, Japanese locale, ISDN Switch type being NTT, JPNP for Numbering Plan are implemented.

The objective of Cisco Unified Communications System Test for Japan is to run a sub-set of system testing that is not covered by Cisco Unified Communications System Test and implement equivalents with Japanese environment such as Japanese OS, localized application, select Cisco Compatible Products, and third party equipment.

In this Cisco Unified Communications System Test release for Japan, the following components are tested.

- Cisco Unified Communications Manager
- Cisco TelePresence Video Communication Server
- Cisco Unified Attendant Console
- Cisco Jabber for iPhone and iPad
- Cisco Jabber for Android
- Cisco Jabber for Windows
- Cisco Jabber for Mac
- Cisco Unified Border Element
- Cisco Unified Survivable Remote Site Telephony
- Cisco Unity Connection
- Cisco Unified CM IM and Presence
- Cisco UC Integration ™ for Microsoft Lync
- Cisco Unified Communications Manager Express
- Cisco TelePresence Multipoint Control Unit
- Cisco TelePresence Management Suite
- Cisco TelePresence Server 7010
- Cisco TelePresence Server on VM
- Cisco TelePresence Conductor
- Cisco Expressway

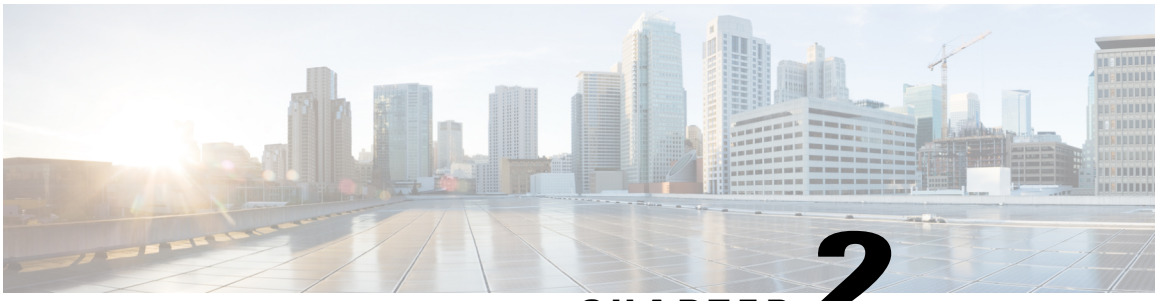
Acronyms

Acronym	Description
AMWI	Audible Message Waiting Indicator
AAR	Automated Alternate Routing
ANAT	Alternate Network Address Translation
ACN	Alternate Contact Number
ACD	Automatic Call Distribution
AGC	Automatic gain control
ASA	Adaptive Security Appliance
ASCII	American Standard Code for Information Interchange
ATA	Analog Telephone Adapter
BAT	Bulk Administration tool
BLF	Busy Lamp Field
CAS	Channel Associated Signaling
CCD	Call Control Discovery
CDR	Call Detail Record
CED	Caller Entered Digits
CFA	Call Forward All
CFB	Call Forward Busy
CFD	Customer Found Defect
CFNA	Call Forward No Answer
CFNC	Call Forward No Coverage
CFUR	Call Forward Unregistered
CJA	Cisco Jabber for Android
CJI	Cisco Jabber for iPhone
CJIP	Cisco Jabber for iPad
CJW	Cisco Jabber For Windows
CLI	Command Line Interface
CLID	Calling Line Identification
CME	Cisco Unified Communications Manager Express
CoW	Clustering over WAN

Acronym	Description
CSF	Client Services Framework
CSS	Calling Search Space
CTI	Computer Telephony Interface
CTI	Computer Telephony Integration
CTL	Certificate Trust List
CUAC	Cisco Unified Attendant Console
CUBE	Cisco Unified Border Element
CUC	Cisco Unity Connection
CUCI-LYNC	Cisco UC Integration ™ for Microsoft LYNC
CUCM	Cisco Unified Communications Manager
DCR	Device and Credential Repository
DHCP	Dynamic Host Configuration Protocol
DID	Direct In-Ward Dialing
DN	Directory Number
DND	Do Not Disturb
DO	Delayed Offer
DPNSS	Digital Private Network Signaling System
DSCP	Differentiated Services Code Point
EM	Extension Mobility
EMCC	Extension Mobility Cross Cluster
EO	Early Offer
FXS	Foreign Exchange Station
GW	Gateway
GUI	Graphical User Interface
HR	Historical Reporting
HA	High Availability
HD	High Definition
HTML	Hyper Text Markup Language
ICT	Inter Cluster Trunk
IM	Instant Messaging
IPPM	IP Phone Messenger

Acronym	Description
ISDN	Integrated Services Digital Network
ITL	Initial Trust List
KEM	Key Expansion Module
LDAP	Light Weight Directory Access Protocol
MCU	Multipoint Control Unit
MGCP	Media Gateway Control Protocol
MOH	Music on hold
MRGL	Media Resource Group List
MWI	Message Waiting Indicator
NLP	Non Linear Processing
NTP	Network Time Protocol
OSD	On Screen Display
POTS	Plain Old Telephony System
PCA	Personal Communication Assistant
PCoIP	PC over IP
PFS	Peer Firmware Sharing
PIN	Personal identification number
PIP	Picture in Picture
PRI	Primary Rate Interface
PRT	Problem Reporting Tool
PSTN	Public Switched Telephone Network
QRT	Quality Report Tool
QSIG	Q-Signaling protocol
RSS	Really Simple Syndication
RTMT	Real Time Monitoring Tool
RTP	Realtime Transport Protocol
SAF	Service Advertisement Framework
SIP	Session Initiation Protocol
SME	Cisco Unified Communications Manager Session Management Edition
SCCP	Skinny Client Control Protocol
SD	Standard Definition

Acronym	Description
SRST	Cisco Unified Survivable Remote Site Telephony
SSL	Secure Socket layer
TAC	Technical Assistant center
TCP	Transmission Control Protocol
TLS	Transport Layer Security
TMS	Cisco Telepresence Management Suite
TPC	Cisco TelePresence Conductor
TRP	Trust Relay Point
TS	Cisco TelePresence Server
TUI	Telephony User Interface
UCS	Unified Computing System
UMG	Unified Messaging Gateway
URI	Uniform Resource Identifier
VCS	Cisco TelePresence Video Communication Server
VGW	Voice Gateway
VoIP	Voice over IP
VPIM	Voice Profile for Instant Messaging
VMN	Voice Mail Notification
VPN	Virtual Private Network
VTS	TelePresence Server on VM
WAN	Wide Area Network
Wi-Fi	Wireless Fidelity
WLC	Wireless LAN Controller

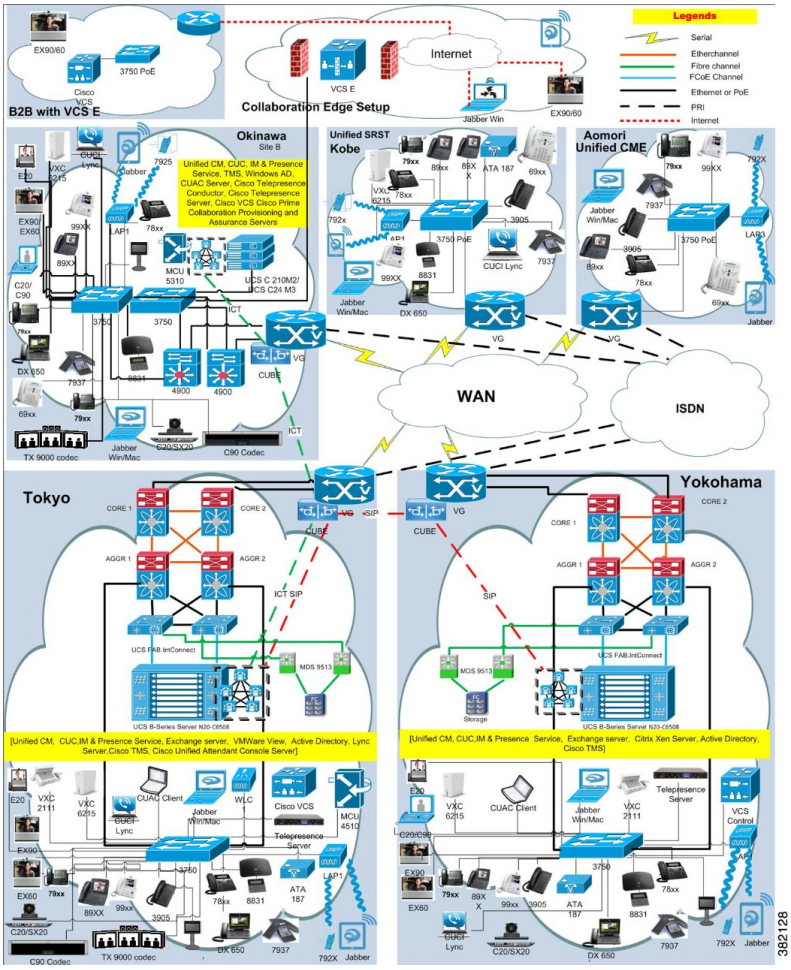


CHAPTER 2

Test Topology and Environment Matrix

Test Topology

Figure 1: Topology in Use



Environment Matrix

Component Matrix			
Applications	**Component**	**Details**	**Version**
Call Control	Cisco Unified Communications Manager		10.0.1.10000-24
		Locale	10.0.1.9902-70(JP)
		Dial Plan	3-1-9.JP
	Cisco Unified Communications Manager Express		10.0(1)
		IOS	15.4(1.24)T0
		Locale	10.0.2.7 (JP)
	Cisco Unified Survivable Remote Site Telephony (SRST)		10.0(1)
		IOS	15.4(1.24)T0
	Cisco Telepresence Video Communication Server (VCS)		X8.1
		Locale	X7.2_LanguagePacks_BETA
Applications	Cisco Unified Attendant Console		10.0.1.10 - 63
	Cisco Unified Communications Manager IM and Presence Service		10.0.1.10000-26
		Locale	10.0.1.1000-1 (JP)
Voice Mail and Unified Messaging	Cisco Unity Connection		10.0.1.10000-24
		Locale	10.0.0.1-1(JP)

Component Matrix			
Applications	**Component**	**Details**	**Version**
End points	Unified IP Phones 6921/41/45/61		9-4-1-3
	Cisco Unified IP Phones 7942/62/75		9-3-1SR4-1
	Cisco Unified IP Phone 7925		1.4(5)
	Cisco Unified IP Phone 8831		9-3-3-5
	Cisco Unified IP Phone 7821/41/61		10.1.1.9
	Cisco Unified IP Phones 8941/8945		9-4-1-8
	Cisco Unified IP Phones 9951/9971		9-4-1-9
	Cisco Desktop Collaboration Experience DX650		10-2-1JBT0-222
	Cisco UC Integration ™ for Lync		9.6.0.621
	EX60 - Cisco TelePresence System EX60		TC7.0.2
	EX90 - Cisco TelePresence System EX90		TC7.0.2
	SX20 - Cisco TelePresence SX20 Quick Set		TC7.0.2
	C20 - Cisco TelePresence System Quick Set C20		TC7.0.2
	Cisco TelePresence System Integrator Package C90		TC7.0.2
	500 - 32 Cisco TelePresence System 500-32		TX6.1.1(50)
	Cisco TX9000 -Cisco TelePresence TX9000		TX6.1.1(50)

Component Matrix			
Applications	**Component**	**Details**	**Version**
Communications Infrastructure	Gateways	IOS	15.4(1)T1
	Cisco Unified Border Element for ISR		10.0.0
	Fabric Interconnect PRIMARY	Cisco UCS 6140	2.1.1a
	Fabric Interconnect SUBORDINATE	Cisco UCS 6140	2.1.1a
	Fabric Cluster	Cisco UCS 6140	2.1.1a
	ESXi host		ESXi 5.1.0
	VCenter Server		ESXi 5.1.0
	MDS Switch	M9500	5.2(2a)
	Cisco Analog Telephone Adaptor	ATA 187	9.2.3
	Cisco 3750 PoE Switch		15.0.2-SE5
TelePresence	Cisco Telepresence Management Suite		14.3.2
	Cisco Telepresence MCU	4510	4.4(3.67)
		5310	4.4(3.67)
		Locale	MCU_4-3_UI_and_audio_JPN.package
	Cisco Telepresence Server on VM		3.1(1.96)
	Cisco Telepresence Conductor		XC2.2.1
	Cisco Telepresence Server 7010		3.1(1.97)

Component Matrix			
Applications	**Component**	**Details**	**Version**
Wireless and Mobility	Wireless LAN Controller 2504		7.6.110.0
	Wireless Access Point 1142		15.2
	Wireless Access Point 35XX		15.2
	Cisco Jabber for Mac		9.2.2.165297
	Cisco Jabber for Windows		9.6.0.17088
	Cisco Jabber for iPhone and iPad		9.6.1.163935
		iPhone 5	Apple iOS 7.0.4
		iPad	Apple iOS 7.0.4
	Cisco Jabber for Android		9.6.0.165769
		Galaxy SII and S4	Android OS 4.2.2
Client	Operating System	Windows 7 -SP1	Windows 7 - SP1 (Japanese)
		Windows 8	Windows 8 (Japanese)
		Mac	10.8
	Browser	IE	IE 9,10 (Supported Japanese language)
		Mozilla	Firefox 27.0.1 (Supported Japanese language)
		Chrome	Chrome 33.0 (Supported Japanese language)
	Microsoft Lync Client		2010
Server	Windows Server		Windows Server 2008 (R2 Enterprise - Japanese)
			Windows Server 2012 (Japanese)
	Exchange Server		2010
	Lync Server		2010

What's New ?

Cisco Unified Attendant Console

Cisco Unified Attendant Console Advanced replaces Cisco Unified Attendant Console Business Edition , Cisco Unified Attendant Console Enterprise Edition , Cisco Unified Attendant Console Premium Edition and consolidates their Functionality into One product .Its powerful call queuing engine helps your operators easily manage a high call volume from a variety of sources. The robust directory can handle up to 100,000 contacts and synchronize directly with Microsoft Active Directory.

Cisco Desktop Collaboration Experience DX650

Mid-sized and enterprise businesses can introduce new ways of collaborating with the DX650. This smart-desk endpoint simplifies workflows and boosts the productivity of employees who work at a desk. Staff can work their way, whether at corporate or branch locations in open office environments, virtual offices or shared workspaces or when teleworking

Cisco Unified IP Conference Phone 8831

The Cisco Unified IP Conference Phone 8831 delivers highly secure, comprehensive, and mission-critical unified communications, with wideband, full-duplex audio performance and flexible accessory options. It is ideal for conference rooms and executive offices within midsize and enterprise businesses supported by Cisco Unified Communications Manager or Cisco Business Edition 6000

Cisco TelePresence Server on VM

In this System Release Cisco TelePresence Server virtualization is supported.

Cisco Expressway Overview

Cisco Expressway will allow connection to the corporate network without the need of VPN. In the Cisco Expressway solution, endpoint registration, call control, provisioning, messaging and presence services are provided by Cisco Unified Communications Manager (Cisco Unified CM). VCS allows users to access the corporate service while working remotely without need for VPN.

Cisco IP Phone 7821/41/61

The Cisco IP Phones 7821/41/61 is easy to deploy and manage, with features such as enhanced flexibility and customization with a field-replaceable bezel option , dedicated fixed keys for common functions such as conference, messaging, and directory, and a two-way navigation button . The Cisco IP Phones 7821/41/61 also supports wideband audio for superior voice quality and offers advanced IP communications features.

Open Caveats

Defect ID	Title
Cisco Unified Communications Manager	
CSCuj76467	Audible Message Waiting indicator is not working in 8831 IP Phone

Defect ID	Title
CSCuj93496	"Enter the number to transfer" option is not found in 8831
CSCum79058	Details showing wrongly at the time of conference in DX650
CSCum81613	FOR DN is displayed wrongly in DX650 when CFA is set.
CSCum81627	Conference members count is not decreasing inDX650when in conference
CSCum82008	No conference timer on 6921/61 phones while pressing Details softkey
CSCum81985	Unable to answer conference in SCCP load after pressing Details softkey
CSCun02402	Phone display language is not updated immediately to phones
CSCun27648	78xx:Phone do not get any ringtone when Default(Ring) is set to phones
Cisco VCS	
CSCul13054	SX20 Quickset Touch-UI Going off State After enabled Extended logging
CSCul01896	Unable to display "Sign In Successful" message in C20 Touch UI for EM user
CSCum74055	Unable to transfer call from EX90 registered in Cisco VCS
CSCun12798	Japanese display ID not coming in SX20 registered in Cisco VCS
CSCun28040	SX20 transmitting video after setting default call rate as 64 kbps
CSCun08178	Complete Transfer option is not coming while transfer using Start new
Cisco TelePresence Conductor	
CSCul04054	8941 IP Phones get reboot after transfer the conference call from EX/SX

CHAPTER 3

Test Results Summary

Cisco Unified Communications Manager

Logical ID	Title	Description	Call Component Flow	Status	Defects
UCJ10PH2SCUCMG661	Place call from 7841 series IP Phone using Speed Dial Button to 9951 IP Phone through CUBE having SIP Trunk with Cluster1 and having SIP Trunk with Cluster2	Verify that user able to make call from 7841 series IP Phone to 99XX IP Phone through CUBE having SIP Trunk with Cluster1 and having SIP Trunk with Cluster2 using Speed Dial Button	7841 IP Phone-> Unified CM 1 -> SIP Trunk->CUBE-> SIP Trunk-> Unified CM 2->99XX IP Phone	Passed	
UCJ10PH2SCUCMG691	Conference from 3 IP phones (7841) via CUBE	To verify that 7841 phone handles the call conference via CUBE successfully	7841 IP Phone1 -> Unified CM1-> SIP trunk -> CUBE -> SIP trunk -> Unified CM2 -> 7841 IP Phone2 -> Unified CM2-> 7841 IP Phone3	Passed	
UCJ10PH2SCUCMG694	Conference call between IP phones(7841) with consultation from IP phone D using SWAP	To verify that 7841 phone handles the call conference with consultation successfully using non 7841 phones	7841 IP Phone1 -> Unified CM1-> 7841 IP Phone2 -> Unified CM1-> IP Phone3	Passed	

Logical ID	Title	Description	Call Component Flow	Status	Defects
UCJ10Ph2SCUCMG718	Display the DN of IP Phone present in Hunt List in Unified IP Phone A	Verify that the Cisco unified IP phone A displays the DN of the Cisco unified IP phone B (in hunt list) which is in other cluster	7841 IP Phone1 -> Unified CM1 -> SIP trunk -> Unified CM2 -> 7841 IP Phone2	Passed	
UCJ10Ph2SCUCMG744	Make a call from 7841 phone A to phone B, Phone B transfers the call to phone C (792X)	Verify that 7841 phones handles that call transfer successfully to 792x phones	7841 IP Phone1 -> Unified CM1 -> 7841 IP Phone2 -> Unified CM1 -> 792x Phone C	Passed	
UCJ10Ph2SCUCMG047	Make call to Unified IP Phone A from Unified IP Phone B where Unified IP Phone B is assigned to a CUCM user via Self-Care portal GUI page(intra)	Verify whether call can be made successfully using Unified IP Phone A to Unified IP Phone B, where the Unified IP Phone B is assigned to the particular CUCM user via Self-care Portal GUI page	Unified IP Phone A -> Unified CM -> Unified IP Phone B	Passed	

Logical ID	Title	Description	Call Component Flow	Status	Defects
UCJ10PH2SCUCMG048	Make call to Unified IP Phone A from Unified IP Phone B where Unified IP Phone B is assigned to a CUCM user via Self-Care portal GUI page(SIP)	Verify whether call can be made successfully using Unified IP Phone A registered in Unified CM 1 to Unified IP Phone B registered in Unified CM 2 via SIP trunk, where the Unified IP Phone B is assigned to the particular CUCM user via Self-care Portal GUI page	Unified IP Phone A -> Unified CM 1 -> SIP Trunk -> Unified CM 2 -> Unified IP Phone B	Passed	
UCJ10PH2SCUCMG240	Call to Manager is handled by Single Assistant in Shared line Mode	Verify whether the calls to manager is handled by single assistant when in shared line mode	Phone C -> Unified CM1 -> Phone A -> Unified CM1 -> Phone B	Passed	
UCJ10PH2SCUCMG245	DND Ringer off on Managers phone	Verify whether the manager can toggle between the DND ringer off and on feature on their phone	Phone A -> Unified CM1 -> Phone B	Passed	

Logical ID	Title	Description	Call Component Flow	Status	Defects
UCJ10Ph2SCUCMG002	Make call between two Unified IP Phones(69xx) to view the call duration in call history(SIP)	Verify whether call duration is shown in call history in 69xx phones when a call is made from Unified IP Phone A registered in Unified CM 1 to Unified IP Phone B registered in Unified CM 2 via SIP trunk.	Unified IP Phone A -> Unified CM 1 -> SIP trunk -> Unified CM 2 -> Unified IP Phone B	Passed	
UCJ10Ph2SCUCMG008	Make call between two Unified IP Phones(69xx) to view the call duration in call history(PSTN)	Verify whether call duration is shown in call history in 69xx phones when a call is made from Unified IP Phone A registered in Unified CM 1 to Unified IP Phone B registered in Unified CM 2 via PSTN trunk.	Unified IP Phone A -> Unified CM 1 -> PSTN -> Unified CM 2 -> Unified IP Phone B	Passed	

Logical ID	Title	Description	Call Component Flow	Status	Defects
UCJ10Ph2SCUCMG009	Transfer call between Unified IP Phones(69xx) to view the call duration in call history(intra)	Verify whether call duration is shown in call history in 69xx phones when a call is made from Unified IP Phone A to Unified IP Phone B inside a single cluster and transferred the call to Unified IP Phone C.	Unified IP Phone A -> Unified CM -> Unified IP Phone B -> Unified CM -> Unified IP Phone C	Passed	
UCJ10Ph2SCUCMG011	Transfer call between Unified IP Phones(69xx) to view the call duration in call history (ICT)	Verify whether call duration is shown in call history in 69xx phones when a call is made from Unified IP Phone A registered in Unified CM 1 to Unified IP Phone B registered in Unified CM 2 via ICT trunk and transfer the call to Unified IP Phone C which is registered in Unified CM 2.	Unified IP Phone A -> Unified CM 1 -> ICT trunk -> Unified CM 2 -> Unified IP Phone B -> Unified CM 2 -> Unified IP Phone C	Passed	

Logical ID	Title	Description	Call Component Flow	Status	Defects
UCJ10Ph2SCUCMS.123	To verify the conversation audio quality when user make a call from Cluster A to Cluster B through SIP trunk.	To verify the conversation audio quality when user make a call from one Cluster to another Cluster through SIP trunk after enabling the PSTN mode for 69xx series.	Unified IP Phone A -> Unified CM 1 -> SIP Trunk -> Unified CM 2 ->Unified IP Phone B	Passed	
UCJ10Ph2SCUCMS.122	To verify the conversation audio quality when user make a call between 69xx IP phone and other type of IP Phone after enabling the PSTN mode option on 69xx IP phone.	Verify the conversation audio quality when user make a call between 69xx and other type of IP Phone after enabling the PSTN mode option	Unified IP Phone A -> Unified CM -> Unified IP Phone B	Passed	
UCJ10Ph2SCUCMS.124	To verify the conversation audio quality when user make a call from Cluster A to Cluster B through ICT trunk.	To verify the conversation audio quality when user make a call from one Cluster to another Cluster through SIP trunk after enabling the PSTN mode for 69xx series.	Unified IP Phone A -> Unified CM 1 -> ICT Trunk -> Unified CM 2 ->Unified IP Phone B	Passed	

Logical ID	Title	Description	Call Component Flow	Status	Defects
UCJ10PH2SCUCMS125	To verify the conversation audio quality when user make a call from one Cluster to another Cluster through H.323.	To verify the conversation audio quality when user make a call from one Cluster to another Cluster through H.323 after enabling the PSTN mode for 69xx series.	Unified IP Phone A -> Unified CM 1 -> H.323 -> CUBE -> H.323-> Unified CM 2 ->Unified IP Phone B	Passed	
UCJ10PH2SCUCMS126	To verify the conversation audio quality when user make a call from one Cluster to another Cluster through PSTN.	To verify the conversation audio quality when user make a call from one Cluster to another Cluster through PSTN after enabling the PSTN mode for 69xx series.	Unified IP Phone A -> Unified CM 1 -> PSTN -> Unified CM 2 ->Unified IP Phone B	Passed	
UCJ10PH2SCUCMG206	The volume change for calls during meet me conference is not auto saved in 69XX phones	Verify whether the volume change for calls during meet me conference is not auto saved in 69XX phones	Phone A -> Unified CM1 -> Phone B -> Unified CM1 -> Phone C	Passed	

Logical ID	Title	Description	Call Component Flow	Status	Defects
UCJ10Ph2SCUCMG209	The volume change during chained conference in 894X phones is saved automatically	Verify whether the volume change during chained conference in 894X phones is saved automatically.	Phone A -> Unified CM1 -> Phone B; Phone A -> Unified CM1 -> Phone C; Phone A -> Unified CM1 -> Phone D	Passed	
UCJ10Ph2SCUCMG223	Video conferencing for phones where firmware shared via PFS	Verify whether the video conference among phones where firmware shared via PFS is successful	Phone A -> Unified CM1 -> Phone B -> Unified CM1 -> Phone C	Passed	
UCJ10Ph2SCUCMG224	Chained conferencing for phones where firmware shared via PFS	Verify whether the chained conference among phones where firmware shared via PFS is successful	Phone A -> Unified CM1 -> Phone B -> Unified CM1 -> Phone C -> Unified CM1 -> Phone D -> Unified CM1 -> Phone E	Passed	
UCJ10Ph2SCUCMG203	The volume change for remote-in-use is saved automatically in 69XX phones	Verify whether the volume change during calls for remote-in-use phone (69XX) is saved automatically	Phone A -> Unified CM1 -> Phone B; Phone D -> Unified CM1 -> Phone C	Passed	

Logical ID	Title	Description	Call Component Flow	Status	Defects
UCJ10Ph2SCUCMG049	Make call to Unified IP Phone A from Unified IP Phone B where Unified IP Phone B is assigned to a CUCM user via Self-Care portal GUI page(ICT)	Verify whether call can be made successfully using Unified IP Phone A registered in Unified CM 1 to Unified IP Phone B registered in Unified CM 2 via ICT trunk, where the Unified IP Phone B is assigned to the particular CUCM user via Self-care Portal GUI page	Unified IP Phone A -> Unified CM 1-> ICT trunk -> Unified CM 2 -> Unified IP Phone B	Passed	
UCJ10Ph2SCUCMG050	Make call to Unified IP Phone A from Unified IP Phone B where Unified IP Phone B is assigned to a CUCM user via Self-Care portal GUI page(PSTN)	Verify whether call can be made successfully using Unified IP Phone A registered in Unified CM 1 to Unified IP Phone B registered in Unified CM 2 via PSTN trunk, where the Unified IP Phone B is assigned to the particular CUCM user via Self-care Portal GUI page	Unified IP Phone A -> Unified CM 1 -> PSTN -> Unified CM 2 -> Unified IP Phone B	Passed	

Logical ID	Title	Description	Call Component Flow	Status	Defects
UCJ10Ph2SCUCMG221	Call to Hunt pilot for phones where firmware shared via PFS	Verify whether the call to hunt pilot number for phones where firmware shared via PFS is successful	Phone D -> Unified CM1 -> Hunt pilot -> Unified CM1 -> Phone A	Passed	
UCJ10Ph2SCUCMG239	Transfer an incoming call to Voice Messaging Service on Managers phone	Verify whether the manager can transfer an incoming call to voice messaging service.	Phone A -> Unified CM1 -> Phone B -> Unified CM1 -> Voice Mail	Passed	
UCJ10Ph2SCUCMG225	Collecting Phone Logs of single SIP phone by setting the profile to Preset	Verify whether the phone logs for single SIP phone are collected from remote log server by setting the profile as preset.	Phone A -> Unified CM1 -> Phone B	Passed	
UCJ10Ph2SCUCMG229	Collecting Phone Logs for single SIP phone by setting the LEVEL options for Log server	Verify whether the phone logs for single SIP phone are collected from remote log server when the LEVEL option is set for Log server	Phone A -> Unified CM1 -> Phone B	Passed	

Logical ID	Title	Description	Call Component Flow	Status	Defects
UCJ10PH2SCUCMG016	Transfer call between Unified IP Phones(69xx) to view the call duration in call history (PSTN)	Verify whether call duration is shown in call history in 69xx phones when a call is made from Unified IP Phone A registered in Unified CM 1 to Unified IP Phone B registered in Unified CM 2 via PSTN trunk and transfer the call to Unified IP Phone C which is registered in Unified CM 2.	Unified IP Phone A -> Unified CM 1 -> PSTN -> Unified CM 2 -> Unified IP Phone B -> Unified CM 2 -> Unified IP Phone C	Passed	

Logical ID	Title	Description	Call Component Flow	Status	Defects
UCJ10Ph2SCUCMG018	Park the call and retrieve in Unified IP Phones(69xx) to view the call duration in call history (SIP)	Verify whether call duration is shown in call history in 69xx phones when a call is made from Unified IP Phone A registered in Unified CM 1 to Unified IP Phone B registered in Unified CM 2 via SIP trunk and park the call in Unified IP Phone B and retrieved the call in Unified IP Phone C which is registered in Unified CM 2.	Unified IP Phone A -> Unified CM 1 -> SIP trunk -> Unified CM 2 -> Unified IP Phone B -> Unified CM 2 -> Unified IP Phone C	Passed	

Logical ID	Title	Description	Call Component Flow	Status	Defects
UCJ10PH2SCUCMG019	Park the call and retrieve in Unified IP Phones(69xx) to view the call duration in call history (ICT)	Verify whether call duration is shown in call history in 69xx phones when a call is made from Unified IP Phone A registered in Unified CM 1 to Unified IP Phone B registered in Unified CM 2 via ICT trunk and park the call in Unified IP Phone B and retrieved the call in Unified IP Phone C which is registered in Unified CM 2.	Unified IP Phone A -> Unified CM 1 -> ICT trunk -> Unified CM 2 -> Unified IP Phone B -> Unified CM 2 -> Unified IP Phone C	Passed	

Logical ID	Title	Description	Call Component Flow	Status	Defects
UCJ10Ph2SCUCMG024	Park the call and retrieve in Unified IP Phones(69xx) to view the call duration in call history (PSTN)	Verify whether call duration is shown in call history in 69xx phones when a call is made from Unified IP Phone A registered in Unified CM 1 to Unified IP Phone B registered in Unified CM 2 via PSTN trunk and park the call in Unified IP Phone B and retrieved the call in Unified IP Phone C which is registered in Unified CM 2.	Unified IP Phone A -> Unified CM 1 -> PSTN -> Unified CM 2 -> Unified IP Phone B -> Unified CM 2 -> Unified IP Phone C	Passed	
UCJ10Ph2SCUCMS101	Log-in allows automatically when user launches the "Cisco Unified CM Administration" web application.	Verify that "Cisco Unified CM Administration" web application should log in without prompting any user credential.	NA	Passed	

Logical ID	Title	Description	Call Component Flow	Status	Defects
UCJ10Ph2SCUCMS102	Log-in allows automatically when user launches the "Cisco Unified Operating System Administration" web application.	Verify that "Cisco Unified Operating System Administration" web application should log in without prompting any user credential.	NA	Passed	
UCJ10Ph2SCUCMG901	Answer the call on 7821 series IP Phone using Auto Answer with Speakerphone	Verify that call on 7821 series IP Phone gets attend automatically using Auto Answer with Speakerphone option when Speakerphone is on in 7821 series IP Phone	Phone A -> Cluster1 -> Phone B	Passed	
UCJ10Ph2SCUCMG903	Forward the calls coming to 7821 series IP Phone using Call Forward All Feature	Verify that calls coming to 7821 series IP Phone forwarding successfully to destination while using Call Forward All Feature	Phone A -> Unified CM1-> Phone B -> Unified CM 1 -> Phone B	Passed	

Logical ID	Title	Description	Call Component Flow	Status	Defects
UCJ10Ph2SCUCMG904	Forward the calls coming to 7821 series IP Phone using Conditional Call Forwarding Feature	Verify that calls coming to 7821 series IP Phone forwarding successfully to destination while using Conditional Call Forwarding Feature	Phone A -> Unified CM 1->Phone B -> Unified CM 1- > Phone C	Passed	
UCJ10Ph2SCUCMG906	Park an inter cluster active call coming through SIP Trunk with Cluster1 and Cluster2 on 7821 series IP Phone	Verify that user able to park an inter cluster active call coming through SIP Trunk with Cluster1 and Cluster2 on 7821 series IP Phone and retrieve the parked call on another IP Phone successfully	7821 IP Phone1->Unified CM1->SIP Trunk->Unified CM2->7821 IP Phone2->9951 IP Phone	Passed	
UCJ10Ph2SCUCMG910	Attend shared line call in 7821 series IP Phone	Verify that user able to attend shared line call in 7821 series IP Phone successfully	7821 IP Phone -> Unified CM-> 7861 IP Phone	Passed	
UCJ10Ph2SCUCMG911	Place call from 7821 series IP Phone using Speed Dial Button	Verify that user able to make call from 7821 series IP Phone to any other IP Phone using Speed Dial Button	7861 IP Phone -> Unified CM -> 9951 IP Phone	Passed	

Logical ID	Title	Description	Call Component Flow	Status	Defects
UCJ10Ph2SCUCMG912	Place call from 7821 series IP Phone using Speed Dial Button to 9951 IP Phone through SIP Trunk with Cluster1 and Cluster2	Verify that user able to make call from 7821 series IP Phone to 9951 IP Phone through SIP Trunk with Cluster1 and Cluster2 using Speed Dial Button	7861 IP Phone->Unified CM1->SIP Trunk->Unified CMr2->9951 IP Phone	Passed	
UCJ10Ph2SCUCMG913	Transfer the call from 7821 series IP Phone to another number	Verify that user able to transfer the call from 7821 series IP Phone to another number successfully	7861 IP Phone1 -> Unified CM1 -> 7861 IP Phone2 -> Unified CM1 -> 9951 IP Phone	Passed	
UCJ10Ph2SCUCMG915	Make a call Form IP Phone A(7821) to IP phone B(7821),call from IP phone C to IP phone A and hear the call waiting tone	To verify that the 7821 phones handles the when second line call waiting tone on single beep mode successfully	1.7861 IP Phone A ->Unified CM-> IP phone B 2.7861 IP phone C->Unified CM->IP Phone A	Passed	
UCJ10Ph2SCUCMG918	IP phones(7821) calls using Forced authorization code	To verify that 7821 phone handles the correctly using FAC before calling	7821 IP Phone1 -> Unified CM -> 7821 IP Phone2	Passed	
UCJ10Ph2SCUCMG919	Make Conference from an IP phones (7821) A, B, C and D	To verify that 7821 phone handles the call conference successfully	7821 IP Phone1 -> Unified CM1-> 7821 IP Phone2 -> Unified CM1-> 7821 IP Phone3 -> Unified CM1 -> 7821 IP Phone4	Passed	

Logical ID	Title	Description	Call Component Flow	Status	Defects
UCJ10Ph2SCUCMG920	Divert calls using IP phones(7821)	Verify that a call is redirected by Immediate Divert to a voice-messaging mail box that is specified in the voice-messaging profile associated with the calling party	7821 IP Phone1 -> Unified CM1 -> 7821 IP Phone2 -> Unified CM1 -> Voicemail	Passed	
UCJ10Ph2SCUCMG921	Activate Do Not Disturb feature on an IP Phone(7821)	Verify that the 7821 IP Phone handles the DND feature in a Cisco Unified IP Phone successfully.	7821 IP Phon2 -> Unified CM1-> 7821 IP Phone1	Passed	
UCJ10Ph2SCUCMG922	Basic IP-to-IP Call Functionality after Logging into Unified Call Manager Assistant	Verify the ability to make a basic IP-to-IP call after logging into the Cisco Unified Call Manager Extension Mobility.	7821 IP Phone1 -> Unified CM1 -> 7821 IP Phone2	Passed	
UCJ10Ph2SCUCMG923	Hold reversion using IP phones(7821)	To verify that the IP phones handles the hold reversion successfully in allocated duration.	7821 IP Phone1 -> Unified CM1 -> 7821 IP Phone2	Passed	

Logical ID	Title	Description	Call Component Flow	Status	Defects
UCJ10Ph2SCUCMG925	Display of connected phone's DN in Cisco unified IP phone A when call made to hunt list present within cluster A	Verify that Cisco unified IP phone A from site A Unified CM makes call to Cisco unified IP phones present in hunt list by dialing hunt pilot number	7821 IP Phone->Unified CM1-> SIP Trunk->Unified CM2->HuntPilot	Passed	
UCJ10Ph2SCUCMG927	Verify line indications status using IP Phones (7821)	Verify that the Cisco unified IP phones(7821) handles the line indications status display successfully	No	Passed	
UCJ10Ph2SCUCMG928	Call made from 7821 Phone to a meet me conference number to join a conference	Verify that the Phone D(7821 Phone) joins a meet me conference call where Phone A, Phone B and Phone C is already in a conference call	7821 IP Phone1 -> Unified CM1 -> 7821 IP Phone2 -> Unified CM1 -> 7821 IP Phone3-> Unified CM1 -> 7821 IP Phone4 -> Meet-me	Passed	
UCJ10Ph2SCUCMG930	Mute : make a call from IP Phone A (7821) to IP Phone B and mute	Verify that IP Phone(7821) handles the mute successfully	7821 IP Phone1 -> Unified CM1-> 7821 IP Phone2	Passed	

Logical ID	Title	Description	Call Component Flow	Status	Defects
UCJ10Ph2SCUCMG934	Make call between 7821 IP phone A to 7925 Phone and vice versa	To verify whether both 7821 IP phone and 7925 phones are registered and attempting a call from 7821 IP Phone to 7925G.	7821 IP Phone1 -> Unified CM1 -> 795 Phone A	Passed	
UCJ10Ph2SCUCMG937	Answer the inter-cluster call on 7861 series IP Phone coming through SIP Trunk with Cluster1 and Cluster2 using Auto Answer with Speakerphone option	Verify that call on 7861 series IP Phone coming through SIP Trunk with Cluster1 and Cluster2 gets attended automatically using Auto Answer with Speakerphone option when Speakerphone on in 7861 series IP Phone	99XX IP Phone->Unified CM2-> SIP Trunk-> Unified CM1->7861 IP Phone	Passed	
UCJ10Ph2SCUCMG939	Forward the inter-cluster calls coming through SIP Trunk with Cluster1 and Cluster2 to 7861 series IP Phone using Call Forward All Feature	Verify that calls coming through SIP Trunk with Cluster1 and Cluster2 to 7861 series IP Phone forwarding successfully to destination while using Call Forward All Feature	9951 IP Phone1-> Unified CM1 ->9951 IP Phone2-> SIP Trunk->Unified CMr2->7861 IP Phone1	Passed	

Logical ID	Title	Description	Call Component Flow	Status	Defects
UCJ10PH2SCUCMG941	Forward the calls coming through SIP Trunk with Cluster1 and Cluster2 to 7861 series IP Phone using Conditional Call Forwarding Feature	Verify that calls coming through SIP Trunk with Cluster1 and Cluster2 to 7861 series IP Phone forwarding successfully to destination while using Conditional Call Forwarding Feature	9951 IP Phone1-> Unified CM1 -> 9951 IP Phone2->Unified CM1-> SIP Trunk->Unified CM2->7861 IP Phone1	Passed	
UCJ10PH2SCUCMG942	Park an active call on 7861 series IP Phone using assisted directed call park	Verify that user able to park an active call on 7861 series IP Phone and retrieve the parked call using Feature button successfully	7861 IP Phone1-> Unified CM->7861 IP Phone2	Passed	
UCJ10PH2SCUCMG946	Place call from 7861 series IP Phone using Speed Dial Button to 9951 IP Phone through ICT Trunk with Cluster1 and Cluster2	Verify that user able to make call from 7861 series IP Phone to 9951 IP Phone through ICT Trunk with Cluster1 and Cluster2 using Speed Dial Button	7861 IP Phone->Unified CM1-> ICT Trunk->Unified CMr2->9951 IP Phone	Passed	

Logical ID	Title	Description	Call Component Flow	Status	Defects
UCJ10Ph2SCUCMG948	Transfer the inter cluster call coming through ICT Trunk with Cluster1 and Cluster2 on 7861 series IP Phone to another number	Verify that user able to transfer the inter cluster call coming through ICT Trunk with Cluster1 and Cluster2 on 7861 series IP Phone to another number successfully	7861 IP Phone1->Unified CM1-> ICT Trunk->Unified CM2->7861 IP Phone2->9951 IP Phone	Passed	
UCJ10Ph2SCUCMG950	Make a call Form IP Phone A(7861) to IP phone B(7861),Continuous call from IP phone C to IP phone A and hear the call waiting tone	To verify that the 7861 phones handles the call waiting tone on single beep amber flash line mode vice versa successfully while making continuous call	1.7861 IP Phone A ->Unified CM-> IP phone B 2.7861 IP phone C->Unified CM-> IP Phone A	Passed	
UCJ10Ph2SCUCMG953	Make call from IP phone A to IP phone B(7861),Make conference to IP Phone C(non-7861)phones	To verify that 7861 phone handles the call conference successfully using non 7861 phones	7861 IP Phone1 -> Unified CM1-> 7861 IP Phone2 -> Unified CM1-> IP Phone3	Passed	

Logical ID	Title	Description	Call Component Flow	Status	Defects
UCJ10Ph2SCUCMG954	Inter-cluster Call with Immediate Divert by Calling Party	Verify that a call is redirected by Immediate Divert to a voice-messaging mail box that is specified in the voice-messaging profile associated with the calling party	7861 IP Phone1 -> Unified CM1 -> SIP trunk -> Unified CM2 -> 7861 IP Phone2 -> Unified CM2 -> Voicemail	Passed	
UCJ10Ph2SCUCMG955	Activate Do Not Disturb feature on an IP Phone(7861),call from IP phone (non 7861)	Verify that the 7861 IP Phone handles the DND feature in a Cisco Unified IP Phone with non 7861 IP phones successfully.	IP Phone1 -> Unified CM1 -> 7861 IP Phone1	Passed	
UCJ10Ph2SCUCMG956	Call Park after Logging into Unified Call Manager Assistant extension mobility	Verify the ability to park a call after logging into Cisco Unified Call Manager Extension Mobility.	7861 IP Phone1 -> Unified CM1 -> 7861 IP Phone2	Passed	
UCJ10Ph2SCUCMG957	Hold reversion using multiple IP phones(7861),set single ring, flashing amber lines	To verify that the IP phones handles the hold reversion successfully as allocated duration.	7861 IP Phone1 -> Unified CM1 -> 7861 IP Phone2; 7861 IP Phone3 -> Unified CM1 -> 7861 IP Phone4	Passed	

Logical ID	Title	Description	Call Component Flow	Status	Defects
UCJ10Ph2SCUCMG959	Display of connected phone's DN in Cisco unified IP phone A when hunt list present use sign in and sign out	Verify that the Cisco unified IP phone A displays the DN of the Cisco unified IP phone (in hunt list) which is in other cluster	7861 IP Phone->Unified CM1-> SIP Trunk->Unified CM2->Hunt Pilot	Passed	
UCJ10Ph2SCUCMG960	Verify line indications status using IP Phones (7861) in shared line	Verify that the Cisco unified IP phones(7861) handles the line indications status display successfully	No	Passed	
UCJ10Ph2SCUCMG962	Call made from multiple Phone to a meet me conference number to join a conference	Verify that the Phones (7861 Phone) and Phone E joins a meet me conference call where Phone A, Phone B and Phone C is already in a meet-me conference call	7861 IP Phone1 -> Unified CM1 -> 7861 IP Phone2 -> Unified CM1 -> 7861 IP Phone3-> Unified CM1 -> 7861 IP Phone4 -> Unified CM1 -> 7861 IP Phone4 -> Meet-me	Passed	
UCJ10Ph2SCUCMG966	Make a call from 7861 phone A to phone B, Call transfer phone C using 792x phones	Verify that 7861 phones handles that call transfer successfully to 792x phones	7861 IP Phone1 -> Unified CM1 -> 7861 IP Phone2 -> Unified CM1 -> 792x Phone A	Passed	

Logical ID	Title	Description	Call Component Flow	Status	Defects
UCJ10S.CUCM.G.219	DX650 - A calls to DX650 - B retrieve the visual voice mail in DX650 - B	Verify that the DX650 - A makes call to DX650 - B not answer the call in DX650 - B , DX650 - A will leave a Visual voice mail to DX650 - B . Retrieve the visual voice mail in DX650 - B by pressing hard key and giving credentials	DX650 - A ->Unified CM 1-> DX650 - B (Unity Connection)	Passed	
UCJ10S.CUCM.G.297	Redial the call on DX650 by pressing the Last Number Redial	Verify whether redial can be made from DX650 to IP Phone successfully by pressing the last Number Redial Icon on the DX650	DX650 -> Unified CM -> IP Phone	Passed	
UCJ10S.CUCM.G.051	Checking for the Wide band ringtone in Cisco Unified Conference IP Phone 8831 during intra-site calls	Verify whether the new Wideband ringtone from the phone firmware is applied successful to Cisco Unified Conference IP Phone 8831 during intra-site calls	Phone A -> Unified CM -> Phone B	Passed	

Logical ID	Title	Description	Call Component Flow	Status	Defects
UCJ10S.CUCM.G.277	DX650 can able to park and retrieve the incoming call from IP Phone	Verify whether DX650-A can park the incoming call from IP Phone and retrieve the parked call in DX650-B	IP Phone A -> Unified CM -> DX650-A -> Unified CM -> DX650-B	Passed	
UCJ10S.CUCM.G.268	DX650 should forward all the incoming calls to the IP Phone when the call forward all is enabled	Verify whether DX650 should forward all the incoming calls to the IP Phone Directory Number in which the call forward all is enabled	IP Phone A -> Unified CM -> DX650 -> Unified CM -> IP Phone B	Passed	
UCJ10S.CUCM.G.614	Check for the prompt display in 8831 during transfer	Verify whether the prompt is displayed to the user while making transfer in 8831	8831 -> Unified CM -> IP Phone A -> Unified CM -> IP Phone B	Failed	CSCuj93496
UCJ10S.CUCM.G.001	Send a voicemail to 8831 IP Phone with in the Cluster	Verify that the Audible Message Waiting Indicator working properly in 8831 IP Phone.	IP Phone A -> Unified CM -> 8831 IP Phone	Failed	CSCuj76467

Logical ID	Title	Description	Call Component Flow	Status	Defects
UCJ10Ph2SCUCMG060	Make a conference call to DX650 to check the details	Verify whether detail is showing correctly in DX650 in conference, when call made from Unified IP Phone A to Unified IP Phone B and conference made from Unified IP Phone B to DX650.	Unified IP Phone A -> Unified CM -> Unified IP Phone B -> Unified CM -> DX650	Failed	CSCum79058
UCJ10Ph2SCUCMG061	Make a call from Unified IP Phone A to DX650 when CFA is given in Unified IP Phone B and check the details in DX650	Verify whether call log is showing correctly in DX650, when call made from Unified IP Phone A to Unified IP Phone B and call goes to DX650, where CFA is set in Unified IP Phone B to Unified IP Phone C and CFA is set in Unified IP Phone C to DX650	Unified IP Phone A -> Unified CM -> Unified IP Phone B -> Unified CM -> Unified IP Phone C -> Unified CM -> DX650	Failed	CSCum81613

Logical ID	Title	Description	Call Component Flow	Status	Defects
UCJ10Ph2SCUCMG062	Make a call from Unified IP Phone A to Unified IP Phone B and make a conference with DX650	Verify whether conference count is decreasing in DX650, when call made from Unified IP Phone A to Unified IP Phone B and make a conference with Unified IP Phone C and DX650, and end the call in Unified IP Phone B.	Unified IP Phone A -> Unified CM -> Unified IP Phone B -> Unified CM -> Unified IP Phone C -> Unified CM -> DX650	Failed	CSCum81627
UCJ10Ph2SCUCMG411	Check for phone display on 69xx phones after making conference	Verify whether the display on 69xx phones are proper after making conference to 69xx and also after pressing the details soft-key	Phone A -> Unified CM1 -> Phone B -> Unified CM1 -> Phone C	Failed	CSCum82008
UCJ10Ph2SCUCMG412	Answer conference call on 69xx phones after pressing details soft-key	Verify whether the user can able to attend the conference call on 69xx phones after pressing the details soft-key	Phone A -> Unified CM1 -> Phone B -> Unified CM1 -> Phone C	Failed	CSCum81985

Logical ID	Title	Description	Call Component Flow	Status	Defects
UCJ10PH2SCUCMG418	Check whether Phone display language setting is updated to phones via Self-Care	Verify whether Phone display language settings are applied to phones immediately when changes are made via Self-Care	NA	Failed	CSCun02402
UCJ10PH2SCUCMG432	Check whether Ring settings are applied to 7841 phones when set via Self-Care portal	Verify whether the Ring settings set to 7841 phones are applied correctly when settings set via Self-Care portal	Phone B -> Unified CM1 -> Phone A; Phone C -> Unified CM1 -> Phone A	Failed	CSCun27648
UCJ10PH2SCUCMSR012	Call pick up group in 894x Phones	Verify whether the call pick up group works when auto pickup is set to true in service parameters	Unified IP Phone A -> Unified CM -> Unified IP Phone B ->Unified CM -> Unified IP Phone C.	Passed	
UCJ10PH2SCUCMSR014	Call made between two 792x Wireless phones connected with WLC	Verify whether there is two way audio available in 792x Wireless phones when connected with the WLC, when call made between 792x Wireless Phones	Wireless Phone A -> Unified CM -> WLC -> Unified CM -> Wireless Phone B	Passed	

Logical ID	Title	Description	Call Component Flow	Status	Defects
UCJ10Ph2SCUCMSR330	792x IP registered with CUCM through wireless network	792x IP Phone should register with CUCM through WLC - AP enabled network setup	NA	Passed	
UCJ10Ph2SCUCMSR331	Make a outgoing call using 792x wireless IP Phone	Make a outgoing call from 792x Wireless IP Phone to other IP phone	Wireless Phone A -> Unified CM -> IP Phone B	Passed	
UCJ10Ph2SCUCMSR332	Incoming call should work on 792x wireless without any issues.	Incoming call should work on 792x wireless without any issues with respect to wireless issues.	Wireless Phone A -> Unified CM -> Wireless Phone B	Passed	
UCJ10Ph2SCUCMSR333	792x phone should not loose the connectivity during the power save mode.	Before 1.4.1 firmware , there is connection loss during the power save mode of 792x phone. The issue has been fixed in latest firmware that should not be reproduced in the latest firmware.	NA	Passed	
UCJ10Ph2SCUCMSR334	After Connection loss , restoring the connection when Phone has the Signal strength	After Connection loss , restoring the connection when Phone has the Signal strength	NA	Passed	

Logical ID	Title	Description	Call Component Flow	Status	Defects
UCJ10S.CUCM.SR.014	Make a call from 792x IP Phone through SIP trunk and check the call History	Verify whether user can view the dialed number in the call history and can able to dialed the same number from history	792x IP Phone -> Unified CM 1 -> SIP Trunk -> Unified CM 2 -> IP Phone A	Passed	
UCJ10S.CUCM.SR.024	Make a call from IP Phone A with SCCP load to any other IP Phone through SIP Trunk and check the call History in the IP Phone	Verify that the user can view the dialed number in the Placed call history of IP Phone	IP Phone A -> Unified CM 1-> SIP Trunk -> Unified CM 2 -> IP Phone A	Passed	
UCJ10S.CUCM.SR.049	Make an inter-cluster call between two Unified CM clusters via SIP Trunk	Verify that the user can made a call from IP Phone A in Unified CM1 cluster to IP Phone B in Unified CM2 cluster via SIP Trunk	IP Phone A -> Unified CM1 -> SIP Trunk -> Unified CM2 -> IP Phone B	Passed	
UCJ10S.CUCM.SR.063	Check the length Fields in RTP packets log trace in 69xx phones by making call from 69xx IP Phone A to IP Phone B	Verify whether the 69xx phones log trace showing the same length field by making call from IP phone A to IP phone B	IP Phone A -> Unified CM -> IP Phone B	Passed	

Logical ID	Title	Description	Call Component Flow	Status	Defects
UCJ10SCUCMSR075	Check the 79XX IP phone menu back soft key is used for going previous menu	Verify whether the 79xx IP phones on going to directories then to missed call menu back soft key is used for going previous menu	NA	Passed	
UCJ10SCUCMSR077	Make a call from IP Phone A to IP Phone B and transfer the call to IP Phone C , make call from IP Phone D to IP Phone B while Transferring	Verify whether that the IP Phone A calls to IP Phone B and dial digits for transferring call from IP Phone B to IP Phone C check the pop disappears when call made to IP Phone B from IP Phone D	IP Phone A --> Unified CM --> IP Phone B (transfer)--> IP Phone C; IP Phone D-->Unified CM --> IP phone B	Passed	
UCJ10SCUCMSR079	Phone Displays "Unknown" when calling number is hidden during Inter-cluster call transfer via PSTN	Verify whether the phone display shows "Unknown" when calling number is hidden during the call transfer via PSTN	Phone A -> Unified CM 1 -> Phone B -> Unified CM1 -> PSTN -> Unified CM2 -> Phone C	Passed	

Logical ID	Title	Description	Call Component Flow	Status	Defects
UCJ10S.CUCM.SR.082	Phone Displays "Unknown" when calling number is hidden during CFNA via PSTN	Verify whether the phone display shows "Unknown" when calling number is hidden when CFNA is enabled to other cluster via PSTN	Phone A -> Unified CM1 -> Phone B -> Unified CM1 -> PSTN -> Unified CM2 -> Phone C	Passed	
UCJ10S.CUCM.SR.085	Phone Displays "Unknown" when calling number is hidden during Call Pickup via PSTN	Verify whether the phone display shows "Unknown" when calling number is hidden during call pickup via PSTN	Phone A -> Unified CM1 -> PSTN -> Unified CM2 -> Phone B -> Unified CM2 -> Phone C	Passed	
UCJ10S.CUCM.SR.093	Phone Displays "Unknown" when calling number is hidden during hold and resume in shared line via PSTN	Verify whether the phone display shows "Unknown" when calling number is hidden during hold and resume in shared line via PSTN	Phone A -> Unified CM1 -> PSTN -> Unified CM2 -> Phone B -> Unified CM2 -> Phone C	Passed	
UCJ10S.CUCM.SR.105	Check the IP Phone status when the Primary Unified CM down	Verify that the IP Phone A register with secondary Unified CM successfully	NA	Passed	

Cisco TelePresence Video Communication Server

Logical ID	Title	Description	Call Component Flow	Status	Defects
UCJ10Ph2S.VCS.G.001	TCP dump logs in Cisco VCS	Verify whether TCP dump logs can be taken along with diagnostic logs in Cisco TelePresence Video Communication Server	NA	Passed	
UCJ10Ph2S.VCS.G.002	Ethernet statistics available in Touch UI of Integrator Package C90	To verify if Ethernet statistics in Touch UI of Cisco TelePresence System Integrator package C90 is working successfully	NA	Passed	
UCJ10Ph2S.VCS.G.003	Password protection of the Touch administrator menu in Integrator Package C90	To Verify if Sign-in with the video system's administrator password to get access to the Administrator menu on Touch 8" of Cisco TelePresence System Integrator Package C90	NA	Passed	

Logical ID	Title	Description	Call Component Flow	Status	Defects
UCJ10Ph2S.VCS.G.004	Presentation will resume automatically within 10 seconds after disconnecting the source while in video conference using Cisco TS 7010	To verify if Presentation source is connected within 10 seconds (e.g. PC) , the presentation will automatically resume in Cisco TelePresence System Integrator Package C90 while in video conference using Cisco TS 7010	Integrator Package C90 and Quickset C20 -> Unified CM->SIP Trunk -> Cisco TS 7010 -> Presentation sharing	Passed	
UCJ10Ph2S.VCS.G.005	Presentation source comes out of standby (e.g. PC) while in video conference using Cisco TS 7010	To verify if Presentation source comes out of standby (e.g. PC) within 10 seconds , the presentation will automatically resume in Cisco TelePresence Integrator Package C90 while in video conference using Cisco TelePresence Server 7010	Integrator Package C90 and Quickset C20 -> Unified CM -> SIP Trunk-> Cisco TS 7010 -> Presentation sharing	Passed	

Logical ID	Title	Description	Call Component Flow	Status	Defects
UCJ10Ph2S.VCS.G.006	Warning displayed when the end port is a lower number than the start port in Integrator Package C90	To Verify if warning displayed when the end port is a lower number than the start port in Cisco TelePresence System Integrator Package C90	NA	Passed	
UCJ10Ph2S.VCS.G.007	Presentation source entered into standby (e.g PC), and comes out of standby within 10 seconds	To verify if presentation source entered into standby (PC), and comes out of standby within 10 seconds, the presentation will automatically be resumed in Cisco TelePresence SX20 QuickSet	SX20 QuickSet -> Unified CM-> QuickSet C20 ->Presentation sharing -> PC entered into standby-> Comes out of standby within 10 seconds -> Presentation resumed	Passed	
UCJ10Ph2S.VCS.G.008	Call forward all to SIP URI in SX20 Quick Set	Verify whether the call forward all can be set to a SIP URI number of Cisco TelePresence System EX90 in Cisco TelePresence SX20 Quick Set registered in Cisco Unified Communications Manager	EX60-> Unified CM -> SX20 Quick Set -> call forward all (SIP URI) -> Unified CM->EX90	Passed	

Logical ID	Title	Description	Call Component Flow	Status	Defects
UCJ10Ph2S.VCS.G.009	Call forward on Unregistered to SIP URI in SX20 Quick Set	Verify whether the call forward on Unregistered can be set to a SIP URI number of Cisco TelePresence System EX90 in Cisco TelePresence SX20 Quick Set registered in Cisco Unified Communications Manager	EX60-> Unified CM -> SX20 Quick Set -> call forward Unregistered(SIP URI)-> Unified CM->EX90	Passed	
UCJ10Ph2S.VCS.G.010	Call forward on Unregistered to EX90 registered in Cisco VCS in SX20 Quick Set	Verify whether the call forward on Unregistered can be set to a Cisco TelePresence System EX90 registered in Cisco TelePresence Video Communications Server in Cisco TelePresence SX20 Quick Set registered in Cisco Unified Communications Manager	EX60 -> Unified CM -> SX20 Quick Set ->call forward Unregistered ->Unified CM- > SIP Trunk-> Cisco VCS->EX90	Passed	
UCJ10Ph2S.VCS.G.011	Join Meet Me conference in SX20 Quick Set	Verify whether the Cisco TelePresence SX20 Quick Set registered in Cisco Unified Communications Manager can join Meet Me conference using Meet Me conference ID.	SX20 Quick Set -> Meet Me conference ID -> Unified CM-> Meet Me Conference	Passed	

Logical ID	Title	Description	Call Component Flow	Status	Defects
UCJ10Ph2SVCSG012	Join MCU 4510 Meet Me conference in SX20 Quick Set	Verify whether the Cisco TelePresence SX20 Quick Set registered in Cisco Unified Communications Manager can join Meet Me conference hosted by Cisco TelePresence MCU 4510 using Meet Me conference ID.	SX20 Quick Set -> Meet Me conference ID-> Unified CM -> MCU 4510 -> Meet Me Conference	Passed	
UCJ10Ph2SVCSG013	Join MCU 4510 Meet Me conference in SX20 Quick Set registered in Cisco VCS	Verify whether the Cisco TelePresence SX20 Quick Set registered in Cisco TelePresence Video Communications Server can join Meet Me conference hosted by Cisco TelePresence MCU 4510 using Meet Me conference ID.	SX20 Quick Set -> Meet Me conference ID->Cisco VCS -> SIP Trunk -> Unified CM->MCU 4510-> Meet Me Conference	Passed	

Logical ID	Title	Description	Call Component Flow	Status	Defects
UCJ10Ph2S.VCS.G.014	Long duration calls between SX20 Quick Set and EX90	Verify whether the Cisco TelePresence SX20 Quick Set and Cisco TelePresence System EX90 registered in Cisco Unified Communications Manager can make a call between them for long duration(1 Hour)	SX20 Quick Set -> Unified CM -> EX90	Passed	
UCJ10Ph2S.VCS.G.015	Long duration calls between SX20 Quick Set and EX90 registered in Cisco VCS	Verify whether the Cisco TelePresence SX20 Quick Set and Cisco TelePresence System EX90 registered in Cisco TelePresence Video Communication Server can make a call between them for long duration(1 Hour)	SX20 Quick Set -> Cisco VCS -> EX90	Passed	

Logical ID	Title	Description	Call Component Flow	Status	Defects
UCJ10Ph2SVCSG016	Long duration audio calls between SX20 Quick Set and 6961	Verify whether the Cisco TelePresence SX20 Quick Set and Cisco Unified IP phone 6961 registered in Cisco Unified Communications Manager can make a call between them for long duration(1 Hour)	SX20 Quick Set -> Unified CM ->6961	Passed	
UCJ10Ph2SVCSG017	Layout Control in TX9000	Verify whether the presentation content can be moved to main screen in Cisco TelePresence TX9000 registered in Cisco Unified Communications Manager when in a call with Cisco TelePresence System EX90 registered in Cisco Unified Communications Manager	TX9000->Unified CM -> EX90-> Presentation Sharing ->Move Presentation to Main Screen	Passed	

Logical ID	Title	Description	Call Component Flow	Status	Defects
UCJ10Ph2S.VCS.G.018	Layout Control in TX9000 via Unified Border Element	Verify whether the presentation content can be moved to main screen in Cisco TelePresence TX9000 registered in Cisco Unified Communications Manager when in a call with Cisco TelePresence System EX90 registered in Cisco Unified Communications Manager via Cisco Unified Border Element	TX9000 -> Unified CM Cluster-1-> SIP Trunk -> Unified Border Element-> SIP Trunk-> Unified CM Cluster-2-> EX90-> Presentation Sharing-> Move Presentation to Main Screen	Passed	
UCJ10Ph2S.VCS.G.019	Disable the Automatic Sharing of Presentation in TX9000 registered in Unified CM	Verify whether the automatic sharing of presentation can be disabled in CLI of Cisco TelePresence TX9000 registered in Cisco Unified Communications Manager	TX9000 -> Unified CM-> EX60-> Presentation Sharing -> Automatic sharing disable	Passed	

Logical ID	Title	Description	Call Component Flow	Status	Defects
UCJ10Ph2SVCSG020	Call forward all to SIP URI in TX9000	Verify whether the call forward all can be set to a SIP URI number of Cisco TelePresence System EX90 in Cisco TelePresence TX9000 registered in Cisco Unified Communications Manager	EX60 -> Unified CM -> TX9000 -> call forward all(SIP URI) -> Unified CM -> EX90	Passed	
UCJ10Ph2SVCSG021	Call forward all to voicemail in TX9000	Verify whether the call forward all can be set to a voicemail number of in Cisco TelePresence TX9000 registered in Cisco Unified Communications Manager	EX60 -> Unified CM -> TX9000-> call forward all->Unified CM-> voicemail	Passed	
UCJ10Ph2SVCSG022	Call forward on Unregistered to SIP URI in TX9000	Verify whether the call forward on Unregistered can be set to a SIP URI number of Cisco TelePresence System EX90 in Cisco TelePresence TX9000 registered in Cisco Unified Communications Manager	EX60 -> Unified CM -> TX9000-> call forward Unregistered(SIP URI)-> Unified CM -> EX90	Passed	

Logical ID	Title	Description	Call Component Flow	Status	Defects
UCJ10Ph2S.VCS.G.023	Call forward on Unregistered to voicemail in TX9000	Verify whether the call forward on Unregistered can be set to a voicemail in Cisco TelePresence TX9000 registered in Cisco Unified Communications Manager	EX60-> Unified CM -> TX9000-> call forward Unregistered-> Unified CM-> voicemail	Passed	
UCJ10Ph2S.VCS.G.024	Join Meet Me conference in TX9000	Verify whether the Cisco TelePresence TX9000 registered in Cisco Unified Communications Manager can join Meet Me conference using Meet Me conference ID.	TX9000-> Meet Me conference ID-> Unified CM-> Meet Me Conference	Passed	
UCJ10Ph2S.VCS.G.025	Join MCU 4510 Meet Me conference in TX9000	Verify whether the Cisco TelePresence TX9000 registered in Cisco Unified Communications Manager can join Meet Me conference hosted by Cisco TelePresence MCU 4510 using Meet Me conference ID.	TX9000-> Meet Me conference ID-> Unified CM-> MCU 4510->Meet Me Conference	Passed	

Logical ID	Title	Description	Call Component Flow	Status	Defects
UCJ10Ph2SVCSG026	Long duration calls between TX9000 and EX90	Verify whether the Cisco TelePresence TX9000 and Cisco TelePresence System EX90 registered in Cisco Unified Communications Manager can make a call between them for long duration(1 Hour)	TX9000 -> Unified CM-> EX90	Passed	
UCJ10Ph2SVCSG027	Long duration calls between TX9000 and EX90 registered in Cisco VCS	Verify whether the Cisco TelePresence TX9000 registered in Cisco Unified Communications Manager can make a call to Cisco TelePresence System EX90 registered in Cisco TelePresence Video Communication Server and stay in the call for 1 hour	TX9000 -> Unified CM -> SIP Trunk-> Cisco VCS-> EX90	Passed	

Logical ID	Title	Description	Call Component Flow	Status	Defects
UCJ10Ph2S.VCS.G.028	Long duration calls between TX9000 and 6961	Verify whether the Cisco TelePresence TX9000 and Cisco Unified IP phone 6961 registered in Cisco Unified Communications Manager can make a call between them for long duration(1 Hour)	TX9000-> Unified CM->6961	Passed	
UCJ10Ph2S.VCS.G.029	Video Call between 500-32 and QuickSet C20 login with EM User	To Verify if video call established between Cisco TelePresence System 500-32 and Cisco TelePresence System QuickSet C20 registered with Cisco Unified Communications Manager where QuickSet C20 login with Extension Mobility User	500-32->Unified CM->QuickSet C20(EM User)	Passed	

Logical ID	Title	Description	Call Component Flow	Status	Defects
UCJ10Ph2S.VCS.G.030	Inter Cluster Call between 500-32 and QuickSet C20 login with EM User	To Verify if inter cluster call established between Cisco TelePresence System 500-32 and Cisco TelePresence System QuickSet C20 registered with Cisco Unified Communications Manager where QuickSet C20 login with Extension Mobility User	500-32-> Unified CM(Cluster 1) -> SIP Trunk-> Unified CM(Cluster 2) -> QuickSet C20(EM User)	Passed	
UCJ10Ph2S.VCS.G.031	Hold and resume the Inter Cluster video call between 500-32 and QuickSet C20 login with EM User	To Verify if user is able to hold and resume the call between Cisco TelePresence System 500-32 and Cisco TelePresence System QuickSet C20 registered with Cisco Unified Communications Manager where QuickSet C20 login with Extension Mobility User	500-32-> Unified CM(Cluster 1) -> SIP Trunk-> Unified CM(Cluster 2) -> QuickSet C20(EM User) -> Hold / Resume	Passed	

Logical ID	Title	Description	Call Component Flow	Status	Defects
UCJ10Ph2S.VCS.G.032	Presentation Sharing between 500-32 and QuickSet C20 login with EM User	To Verify if user is able to share the presentation between Cisco TelePresence System 500-32 and Cisco TelePresence System QuickSet C20 registered with Cisco Unified Communications Manager where QuickSet C20 login with Extension Mobility User	500-32 -> Unified CM(Cluster 1) -> SIP Trunk-> Unified CM(Cluster 2)-> QuickSet C20(EM User) -> Presentation Sharing	Passed	
UCJ10Ph2S.VCS.G.033	Conference between 500-32 , QuickSet C20 (login with EM User) and Integrator Package C90	To verify if video conference established between Cisco TelePresence System 500-32 , Cisco TelePresence System Integrator Package C90 and Cisco TelePresence System QuickSet C20(login with Extension Mobility) registered with Cisco Unified Communications Manager	500-32 -> Unified CM -> Integrator Package C90 -> QuickSet C20 (EM User)-> Conference	Passed	

Logical ID	Title	Description	Call Component Flow	Status	Defects
UCJ10Ph2SVCSG034	Hold and resume the conference between 500-32 , QuickSet C20 (login with EM User) and Integrator Package C90	To verify if hold and resume the conference between Cisco TelePresence System 500-32 , Cisco TelePresence System Integrator Package C90 and Cisco TelePresence System QuickSet C20(login with Extension Mobility) registered with Cisco Unified Communications Manager	500-32-> Unified CM->Integrator Package C90 -> QuickSet C20 (EM User)-> Conference -> Hold / Resume	Passed	
UCJ10Ph2SVCSG035	Presentation Sharing between 500-32 , QuickSet C20 (login with EM User) and Integrator Package C90	To verify if user is able to share the presentation between Cisco TelePresence System 500-32 , Cisco TelePresence System Integrator Package C90 and Cisco TelePresence System QuickSet C20(login with Extension Mobility) registered with Cisco Unified Communications Manager	500-32-> Unified CM-> Integrator Package C90-> QuickSet C20 (EM User)-> Conference-> Presentation Share	Passed	

Logical ID	Title	Description	Call Component Flow	Status	Defects
UCJ10Ph2S.VCS.G.036	Call Forward All in 500-32 registered with Unified CM	To Verify if video call established between Cisco TelePresence System 500-32 and Cisco TelePresence System QuickSet C20 registered with Cisco Unified Communications Manager and set call forward all in 500-32 to Cisco TelePresence SX20 QuickSet	QuickSet C20-> Unified CM-> 500-32-> CFA-> SX20 QuickSet	Passed	
UCJ10Ph2S.VCS.G.037	Inter cluster call forward all in 500-32 registered with Unified CM	To Verify if inter cluster video call between Cisco TelePresence System 500-32 and Cisco TelePresence System QuickSet C20 registered with Cisco Unified Communications Manager and set call forward all in 500-32 to Cisco TelePresence SX20 QuickSet	QuickSet C20-> Unified CM(Cluster 1)-> 500-32 ->CFA-> Unified CM(Cluster 1)->SIP Trunk->Unified CM(Cluster 2) -> SX20 QuickSet	Passed	

Logical ID	Title	Description	Call Component Flow	Status	Defects
UCJ10Ph2S.VCS.G.038	Call forward all to voicemail in 500-32 registered with Unified CM	To Verify if inter cluster video call between Cisco TelePresence System 500-32 registered with Cisco Unified Communications Manager (Cluster 2) and Cisco TelePresence System QuickSet C20 registered with Cisco Unified Communications Manager(Cluster 1) and set call forward all in 500-32 to voicemail	QuickSet C20-> Unified CM (Cluster 1)-> SIP Trunk->Unified CM (Cluster 2) ->500-32 -> CFA-> voicemail	Passed	

Logical ID	Title	Description	Call Component Flow	Status	Defects
UCJ10Ph2S.VCS.G.039	Call Forward All in 500-32 to SX20 Quickset	To Verify if video call established between Cisco TelePresence System 500-32 and Cisco TelePresence System QuickSet C20 registered with Cisco Unified Communications Manager and set call forward all in 500-32 to Cisco TelePresence SX20 QuickSet where Cisco TelePresence System QuickSet C20 login with Extension Mobility User	QuickSet C20 (EM User)-> Unified CM ->500-32 -> CFA-> SX20 QuickSet	Passed	
UCJ10Ph2S.VCS.G.040	Inter cluster Call between 500-32 and DX650	To Verify if inter cluster call established between Cisco TelePresence System 500-32 and Cisco Desktop Collaboration Experience DX650 registered with Cisco Unified Communications Manager	500-32->Unified CM(Cluster 1)->SIP Trunk->Unified CM(Cluster 2)-> DX650	Passed	

Logical ID	Title	Description	Call Component Flow	Status	Defects
UCJ10Ph2SVCSG041	Hold and resume the call between 500-32 and DX650	To Verify if hold and resume the call between Cisco TelePresence System 500-32 and Cisco Desktop Collaboration Experience DX650 registered with Cisco Unified Communications Manager	500-32->Unified CM(Cluster 1) -> SIP Trunk -> Unified CM(Cluster 2) -> DX650 -> Hold / Resume	Passed	
UCJ10Ph2SVCSG042	Presentation Sharing between 500-32 and DX650	To Verify if presentation sharing between Cisco TelePresence System 500-32 and Cisco Desktop Collaboration Experience DX650 registered with Cisco Unified Communications Manager	500-32->Unified CM(Cluster 1)->SIP Trunk-> Unified CM(Cluster 2) -> DX650 -> Presentation Sharing	Passed	
UCJ10Ph2SVCSG043	Audio call between 500-32 and 8831	To Verify if audio call established between Cisco TelePresence System 500-32 and Cisco Unified IP Conference Station 8831 registered with Cisco Unified Communications Manager	500-32 ->Unified CM->8831	Passed	

Logical ID	Title	Description	Call Component Flow	Status	Defects
UCJ10Ph2S.VCS.G.044	Hold and resume the audio call between 500-32 and 8831	To Verify if hold and resume the audio call between Cisco TelePresence System 500-32 and Cisco Unified IP Conference station 8831 registered with Cisco Unified Communications Manager	500-32-> Unified CM-> 8831-> Hold /Resume	Passed	
UCJ10Ph2S.VCS.G.045	Audio Conference in 500-32 initiated by 8831	To Verify if audio conference between Cisco TelePresence System 500-32 , Cisco Unified IP Conference station 8831 and Cisco TelePresence System SX20 QuickSet registered with Cisco Unified Communications Manager	500-32-> Unified CM -> 8831-> SX20 QuickSet -> Audio Conference	Passed	

Logical ID	Title	Description	Call Component Flow	Status	Defects
UCJ10Ph2S.VCS.G.046	Transfer a call from EX90 registered in Cisco VCS to EX60 registered in Unified CM	Verify whether the call can be transferred from Cisco TelePresence System EX90 registered in Cisco TelePresence Video Communication Server to Cisco TelePresence System EX60 registered in Cisco Unified Communications Manager	EX90-> Cisco VCS-> EX90(1) -> Transfer -> Cisco VCS->SIP Trunk-> Unified CM->EX60	Failed	CSCum74055
UCJ10Ph2S.VCS.G.049	Japanese display ID in SX20 Quick Set registered in Cisco VCS	Verify whether Japanese display ID set for Cisco TelePresence System EX90 registered in Cisco Unified Communications Manager is reflected in Cisco TelePresence SX20 Quick Set registered in Cisco TelePresence Video Communication Server	EX90-> Unified CM->SIP Trunk-> Cisco VCS-> SX20 Quick Set	Failed	CSCun12798

Logical ID	Title	Description	Call Component Flow	Status	Defects
UCJ10Ph2S.VCS.G.050	SX20 Quick Set transmits audio after setting default call rate as 64 kbps	Verify whether SX20 Quick Set registered in Cisco Unified Communications Manager transmits audio when the bandwidth is set as 64 Kbps for the call to Cisco Telepresence System EX90	SX20 Quick Set(64 Kbps) ->Unified CM -> EX90	Failed	CSCun28040
UCJ10Ph2S.VCS.G.051	Make a call between EX60 and EX90 , transfer a call from EX60 to C90	To Verify that user can able to transfer a call from Cisco TelePresence System EX60 to Cisco TelePresence System Integrator Package C90	EX60-> Cisco VCS-> EX90 ; EX90-> transfer-> C90	Failed	CSCun08178
UCJ10S.VCS.G.096	Enable Extended Logging in SX20 Quickset Web-GUI	To verify that user can able to enable extended logging option in SX20 Quickset Web GUI	NA	Failed	CSCul13054

Logical ID	Title	Description	Call Component Flow	Status	Defects
UCJ10S.VCS.G.101	"Sign in Successful " message displayed in QuickSet C20 for EM User	To verify if "Sign in successful " message is displayed in the Touch UI of Cisco TelePresence System QuickSet C20 after login with Extension Mobility User where Extension Mobility user locale is set to Japanese language	NA	Failed	CSCul01896

Cisco Jabber for iPhone and iPad

Logical ID	Title	Description	Call Component Flow	Status	Defects
UCJ10Ph2S.CJI.G.008	Make an intra cluster call from Jabber for iPhone to Jabber for iPad and conference to the Jabber for Android user by using of email id (URI dialing).	Verify that Jabber for iPhone should be in conference with other Jabber for Android and Jabber for iPad	CJI --> Unified CM --> CJiPad --> Unified CM -> CJA	Passed	
UCJ10Ph2S.CJI.G.010	Make an intra cluster call from Jabber for iPhone to Jabber for Windows in Search by using of email id (URI Dialing)	Verify that Jabber for iPhone should call to Jabber for Windows by using of URI dialing.	CJI-->Unified CM-->CJW	Passed	

Logical ID	Title	Description	Call Component Flow	Status	Defects
UCJ10Ph2S.CJI.G.002	Make an inter cluster call from Jabber for iPhone to Jabber for Windows via SIP Trunk.	Verify that Jabber for iPhone should display the secure key icon after connected to Jabber for Windows	CJI --> Unified CM1--> SIP Trunk--> Unified CM2 --> CJW.	Passed	
UCJ10Ph2S.CJI.G.004	Make an inter cluster call from Jabber for iPhone to 9971 IP Phone via Unified Border Element through SIP Trunk.	Verify that Jabber for iPhone should display the secure key icon after connected to 9971 IP Phone.	CJI-->Unified CM1-->SIP Trunk-->Unified CM2-->9971 IP Phone	Passed	
UCJ10Ph2S.CJI.G.007	Make an intra cluster call from Jabber for iPad to Jabber for iPhone and Transfer the call from Jabber for iPhone to Jabber for Android user by using of email id (URI dialing).	Verify that Jabber for iPhone should Transfer the call to Jabber for Android by using of email id(URI dialing)	CJ iPad --> Unified CM --->CJI -->Unified CM -->CJA.	Passed	
UCJ10Ph2S.CJI.G.011	Make an intra cluster call from Jabber for Android to Jabber for iPhone and make conference from Jabber for iPhone to Jabber for Windows by using of email id(URI dialing).	Verify that Jabber for iPhone should make conference to Jabber for Windows by using of email id(URI dialing) and conference with Jabber for Android and Windows.	CJA --> Unified CM --> CJI --> Unified CM --> CJW	Passed	

Logical ID	Title	Description	Call Component Flow	Status	Defects
UCJ10Ph2S.CJI.G.005	Make an inter cluster call from Jabber for iPhone to Jabber for Windows via ICT Trunk.	Verify that Jabber for iPhone should display the secure key icon after connected to the Jabber for Windows	CJI-->Unified CM1-->ICT Trunk--> Unified CM2 - > CJW.	Passed	
UCJ10Ph2S.CJI.G.006	Make an inter cluster call from Jabber for Windows to Jabber for iPhone and make conference from Jabber for iPhone to 9971 IP Phone via Unified Border Element through SIP Trunk.	Verify that Jabber for iPhone should conference with 9971 IP Phone and Jabber for Windows and secure key icon should display in Jabber for iPhone.	CJW --> Unified CM1 --> SIP Trunk -> CUBE -> SIP trunk --> Unified CM2 --> CJI --> Unified CM2 --> CJW.	Passed	
UCJ10Ph2S.CJI.SR.002	Jabber for iPhone couldn't able to search the LDAP when transferring the call.	Verify that Jabber for iPhone should search the contacts from LDAP when transferring the call to Jabber for Android.	CJI ---> Unified CM ---> CJA	Passed	

Logical ID	Title	Description	Call Component Flow	Status	Defects
UCJ10Ph2S.CJiPad.G.002	Make an inter cluster call from Jabber for iPad to 9971 IP Phone and Conference the call to Jabber for Android via SIP Trunk.	Verify that Jabber for iPad should display the secure key icon after Conference with other 9971 IP Phone and Jabber for Android.	CJiPad --> Unified CM1 --> SIP Trunk --> Unified CM2 --> 9971 IP Phone --> Unified CM2 ->CJA	Passed	

Logical ID	Title	Description	Call Component Flow	Status	Defects
UCJ10Ph2S.CJiPad.G.005	Make an inter cluster call from Jabber for iPad to 9971 IP Phone and Conference the call to Quickset C20 via Unified Border Element through ICT Trunk	Verify that Jabber for iPad should display the secure Key icon after Conference with other 9971 IP Phone and Quickset C20.	CJiPad-->Unified CM1-->ICT Trunk-->CUBE-->ICT Trunk-->Unified CM2-->9971 IP Phone-->Unified CM1 -> Quickset C20	Passed	
UCJ10Ph2S.CJiPad.G.011	Make an intra cluster call from Jabber for iPhone to Jabber For iPad by using of email id (URI Dialing) and Transfer the call from Jabber for iPad to Jabber for Android by using of email id.	Verify that Jabber for iPad should transfer the call to Jabber for Android by using of email id(URI Dialing)	CJI--> Unified CM --> CJiPad-->Unified CM -> CJA	Passed	
UCJ10Ph2S.CJiPad.G.012	Make an intra cluster call from Jabber for iPad to Jabber for Android by using email id (URI Dialing) in Favorites	Verify that Jabber for iPad should call to Jabber for Android by using of email id (URI Dialing)	CJiPad-->Unified CM-->CJA	Passed	
UCJ10Ph2S.CJiPad.G.003	Make an intra cluster call from Jabber for iPad to IP Phone 9951.	Verify that Jabber for iPad should display the secure key icon after connected to 9971 IP Phone.	CJiPad-->Unified CM-->9951 IP Phone	Passed	

Logical ID	Title	Description	Call Component Flow	Status	Defects
UCJ10Ph2S.CJiPad.G.006	Make an inter cluster call from Jabber for iPad to 9971 IP Phone and Transfer the call from 9971 IP Phone to Quickset C20 via Unified Border Element through ICT Trunk	Verify that Jabber for iPad should display the secure key icon after transferred the call to Quickset C20.	CJiPad --> Unified CM1 --> SIP Trunk --> CUBE --> ICT Trunk--> Unified CM2 --> 9971 IP Phone ---> Unified CM2 -> Quickset C20.	Passed	
UCJ10Ph2S.CJiPad.G.001	Make an intra-cluster call from Jabber for iPad to Jabber for Android by using of email id (URI dialing).	Verify that Jabber for iPad should display the secure key icon after connected to Jabber for Android.	CJiPad-->Unified CM-->CJA	Passed	
UCJ10Ph2S.CJiPadSR.001	Make an intra cluster call from Jabber for iPad to 6941 IP Phone after upgrade IOS 7 on iPad.	Verify that Jabber for iPad should call to 6941 IP Phone and should work two-ways audio calls after upgrade IOS 7.	CJiPad-->Unified CM-->6941 IP Phone.	Passed	

Cisco Jabber for Android

Logical ID	Title	Description	Call Component Flow	Status	Defects
UCJ10Ph2S.CJA.G.004	Make an intra cluster call from Jabber for Android to Jabber for iPhone by using of email id(URI dialing) in call history.	Verify that Jabber for Android should call to Jabber for iPhone user by using of email id.	CJA-->Unified CM-->CJI.	Passed	
UCJ10Ph2S.CJA.G.005	Make an intra cluster call from Jabber for Android to Jabber for iPad by using of email id (URI dialing) in chat window.	Verify that Jabber for Android should call to Jabber for iPad user by using of email id.	CJA-->Unified CM-->CJ iPad.	Passed	
UCJ10Ph2S.CJA.G.016	Make an inter cluster call from Jabber for Android to 8831 IP Phone and transfer the call to jabber for iPad from 8831 IP Phone via SIP Trunk	Verify that Jabber for Android should display the secure key icon after transferred the call to jabber for iPad	CJA-->Unified CM1-->SIP Trunk--> Unified CM2-->IP Phone 8831--> Unified CM2-->CJ iPad.	Passed	
UCJ10Ph2S.CJA.G.018	Make an inter cluster call from Jabber for Android to Jabber for iPad and transfer the call to Jabber for iPhone from jabber for iPad via Unified Border Element through SIP Trunk	Verify that Jabber for Android should display the secure key icon after transferred the call to Jabber for iPhone	CJA--> Unified CM1-->SIP Trunk--> CUBE-->SIP Trunk--> Unified CM2--> CJ iPad--> Unified CM2 -->CJI	Passed	

Logical ID	Title	Description	Call Component Flow	Status	Defects
UCJ10Ph2S.CJA.G.008	Make an intra cluster call from Jabber for iPhone to Jabber for Android by using of email id(URI dialing) in Contact Profile and Transfer the call from Jabber for Android to Jabber for iPad.	Verify that Jabber for Android should transfer the call to Jabber for iPad by using of email id(URI dialing).	CJI--> Unified CM--> CJA --> Unified CM -->CJ iPad	Passed	
UCJ10Ph2S.CJA.G.015	Make an intra-cluster call from Jabber for Android to Jabber for iPhone	Verify that Jabber for Android should display the secure key icon after connected to the Jabber for iPhone	CJA-->Unified CM-->CJI	Passed	
UCJ10Ph2S.CJA.G.017	Make an inter-cluster call from Jabber for Android to Jabber for iPad via SIP Trunk	Verify that Jabber for Android should display the secure key icon after connected to the Jabber for iPad	CJA--> Unified CM1-->SIP Trunk--> Unified CM2 - > CJ iPad	Passed	

Cisco Jabber for Windows

Logical ID	Title	Description	Call Component Flow	Status	Defects
UCJ10Ph2S.CJW.G.022	By entering number in dock and press "enter" for making call from CJW	Verify that after entering number in dock and then by pressing "enter" check whether call gets established successfully from CJW	CJW A -> Unified CM1 -> SIP trunk -> Unified CM 2 -> Phone A	Passed	
UCJ10Ph2S.CJW.G.160	Initiate call from CJW A to CJW B using URI dialing	Verify that call made from Cisco Jabber for Windows A to Cisco Jabber for Windows B by dialing the URI successfully	CJW A -> Unified CM -> CJW B	Passed	
UCJ10Ph2S.CJW.G.161	CJW B receives incoming call from CJW A using URI dialing and CJW B transfers the call	Verify that Cisco Jabber for Windows A makes call to cisco Jabber for Windows B using URI dialing and Jabber for Windows B transfers the call successfully	CJW A -> Unified CM -> CJW B -> Unified CM -> Phone C	Passed	
UCJ10Ph2S.CJW.G.162	Call forward in CJW B and CJW A makes call to CJW B using URI dialing and IP phone C receives call	Verify that call forward set in CJW B successfully and Cisco Jabber for Windows A makes call to Cisco Jabber for Windows B using URI dialing and IP phone C attends forwarded call	CJW A -> Unified CM -> CJW B -> Unified CM -> Phone C	Passed	

Logical ID	Title	Description	Call Component Flow	Status	Defects
UCJ10Ph2S.CJW.G.026	Group chat invite in CJW	Verify that Cisco Jabber for Windows B and C receives the Group chat invite notification during group chat successfully	NA	Passed	
UCJ10Ph2S.CJW.G.028	Group chat invite sent to user which is already in Chat	Verify that Cisco Jabber for Windows B which is in chat with CJW A receives the Group chat invite notification from Cisco Jabber for Windows A during group chat successfully	NA	Passed	
UCJ10Ph2S.CJW.G.031	Chat history save in Cisco Jabber for Windows	Verify that the chat content of cisco Jabber for Windows are saved successfully	NA	Passed	
UCJ10Ph2S.CJW.G.032	"Sign in as different user" prompt while logging different users present within cluster	Verify that "Sign in as different user" prompt appears when user 2 tries to login to the CJW once the user 1 gets sign out from CJW	NA	Passed	

Logical ID	Title	Description	Call Component Flow	Status	Defects
UCJ10Ph2S.CJW.G.034	Multiple user login to CJW and check prompt status	Verify the prompt "Sign in as different user" appears when multiple users login to the Cisco Jabber for Windows after user 1 gets sign out of CJW	NA	Passed	
UCJ10Ph2S.CJW.SR.002	Call Transfer from CJW A to CJW B in Unified CM 1 via SIP Trunk	Verify that CJW A in Unified CM 2 successfully transfers the call to CJW B in Unified CM 1 via SIP trunk	Phone A -> Unified CM -> CJW A ; CJW A -> Unified CM1 -> SIP Trunk -> Unified CM2 -> CJW B	Passed	
UCJ10Ph2S.CJW.SR.007	Transfer of file between two Cisco Jabber for Windows	Verify that Cisco Jabber for Windows successfully transfer a file to other Cisco Jabber for Windows through chat	NA	Passed	
UCJ10Ph2S.CJW.SR.003	Validate display of video after complete transfer in CJW	Verify that video call is working in CJW and video is displayed in both CJW A and C successfully after complete transfer made by CJW B	CJW A -> Unified CM -> CJW B ; CJW B ->Unified CM -> CJW C	Passed	

Logical ID	Title	Description	Call Component Flow	Status	Defects
UCJ10Ph2S.CJW.SR.005	Validate video call establishment between CJW's when call is transferred via CUBE	Verify that video call is established successfully between CJW A in Unified CM 2 and CJW C in Unified CM 1 when complete transfer made by CJW B via CUBE	CJW A -> Unified CM -> CJW B ; CJW B -> Unified CM1-> H.323 -> CUBE -> H.323 -> Unified CM2 -> CJW C	Passed	
UCJ10Ph2S.CJW.SR.009	Connection settings in Cisco Jabber for Windows	Verify that in Jabber for Windows there is no change in connection settings after the restart of PC	NA	Passed	
UCJ10S.CJW.SR.011	Jabber remains running in system if Jabber exit in Windows 7	Verify that Jabber remains running in Windows 7 Machine check in Task Manager	CJW A---> Cluster ---> CJW B	Passed	
UCJ10S.CJW.SR.015	Jabber remains running in system if Jabber exit in Windows 8 Task Manager	Verify that Jabber remains running in Windows 8 Machine check in Task Manager.	CJW A--->Cluster --->CJW B	Passed	
UCJ10S.CJW.SR.016	Jabber remains running in system if Jabber exit in Windows 8 Status Bar	Verify that jabber remains running in windows 8 Machine check in Status bar.	CJW A--->Cluster ---> CJW B	Passed	

Logical ID	Title	Description	Call Component Flow	Status	Defects
UCJ10S.CJW.SR.018	Make Call to Voicemail in Desk phone mode	Verify that Jabber for Windows invoke to voicemail in desk phone mode after assigning the voicemail account to the user	CJW -> CTI -> Unified IP Phone -> Unity Connection	Passed	
UCJ10S.CJW.SR.021	Make a call from Unified IP Phone 9971 to Jabber for Windows when call forward all number is set to Voicemail	Verify that Jabber for Windows forwards the incoming call to voicemail	Unified IP Phone A-> Cluster -> CJW -> Unity Connection	Passed	
UCJ10S.CJW.SR.022	Call Voicemail button should not be clickable if no voicemail profile is assigned to the user.	Verify that Call voicemail button in Jabber for Windows should not be in enabled mode if no voicemail profile is assigned to the user.	NA	Passed	

Cisco Jabber for Mac

Logical ID	Title	Description	Call Component Flow	Status	Defects
UCJ10Ph2S.CJM.G.001	Video Call from Jabber for Mac to Jabber for iPhone via SIP Trunk	Verify that Cisco Jabber for Mac Client can able to do Video Call to Cisco Jabber for iPhone via SIP trunk in different cluster	Jabber for Mac -> Unified CM1-> SIP trunk->Unified CM2-> Jabber for iPhone	Passed	
UCJ10Ph2S.CJM.G.002	Chat history until logout when Jabber for Mac do Instant Messaging with Jabber for iPhone in same cluster	Verify that Jabber for Mac do Instant Messaging with Jabber for iPhone in same cluster and after logout from Jabber for Mac, Chat messages should not be there in the history	Jabber for Mac -> Unified CM1 -> Jabber for iPhone	Passed	
UCJ10Ph2S.CJM.G.003	Chat history when Jabber for Mac close the chat box after Instant Messaging with Jabber for iPhone in different cluster	Verify that when Jabber for Mac do Instant messaging with Jabber for iPhone in different cluster and close the chat box, Chat messages should be there in the history	Jabber for Mac-> Unified CM1-> SIP trunk-> Unified CM2 -> Jabber for iPhone	Passed	

Logical ID	Title	Description	Call Component Flow	Status	Defects
UCJ10Ph2S.CJM.G.004	Self-Video window can be enabled/disabled in call preferences when Jabber for Mac call connected with Jabber for iPhone via PSTN	Verify that Self-Video window can be enabled/disabled in call preferences when Jabber for Mac call connected with Jabber for iPhone via PSTN	Jabber for Mac -> Unified CM1-> PSTN-> Unified CM2-> Jabber for iPhone	Passed	
UCJ10Ph2S.CJM.G.005	Jabber for Mac do Group chat in same cluster	Verify that when Jabber for Mac do Group chat with other client in same cluster, Group chat invitation does not contain chat topic	Jabber for Mac -> Unified CM1-> Jabber Clients	Passed	
UCJ10Ph2S.CJM.G.006	In Jabber for Mac Chat history, input one name to filter out the chat history with the User in different cluster.	Verify that in Jabber for Mac Chat history, User can able to input one name to filter out the chat history with the User in different cluster.	Jabber for Mac -> Unified CM1-> SIP trunk-> Unified CM2-> Jabber Clients	Passed	
UCJ10Ph2S.CJM.G.007	Always Start Calls with Video in Jabber for Mac when Jabber for iPhone call via ICT Trunk	Verify that Cisco Jabber for Mac Client starts calls with Video when Cisco Jabber for iPhone make call via ICT trunk in different cluster	Jabber for iPhone -> Unified CM1-> ICT trunk -> Unified CM2 ->Jabber for Mac	Passed	

Logical ID	Title	Description	Call Component Flow	Status	Defects
UCJ10Ph2S.CJM.G.008	Video Conferencing from Jabber for Mac to Jabber for iPhone and Jabber for Windows in same cluster	Verify that Jabber for Mac can do Video Conferencing to Jabber for iPhone and Jabber for Windows in same cluster	Jabber for Mac->Unified CM1 -> Jabber for iPhone->Unified CM1->Jabber for Windows	Passed	

Cisco Unified Border Element

Logical ID	Title	Description	Call Component Flow	Status	Defects
UCJ10Ph2SCUBE.G.120	Call made from 8831 IP phone to IP phone B via CUBE through H.323 to H.323 interworking	Verify that 8831 IP phone successfully makes call to IP phone B via CUBE through H.323 to H.323 interworking	8831 -> Unified CM1-> H.323 -> CUBE -> H.323 -> Unified CM2 -> Phone B	Passed	
UCJ10Ph2SCUBE.G.121	Retrieval of parked call by IP phone C when call initiated via CUBE	Verify that IP phone A makes call via CUBE through H.323 to H.323 interworking to IP phone B and IP phone B answers and parks the call successfully and IP phone C retrieves the parked call	Phone A -> Unified CM1 -> H.323 -> CUBE -> H.323 -> Unified CM2 -> Phone B -> Unified CM2 -> Phone C	Passed	

Logical ID	Title	Description	Call Component Flow	Status	Defects
UCJ10Ph2SCUBEG122	Call made to Hunt pilot number via CUBE	Verify that IP phone A makes call via CUBE through H.323 to H.323 interworking to hunt pilot number and IP phone B present in hunt group answers the call successfully	Phone A -> Unified CM1 -> H.323 -> CUBE -> H.323 -> Unified CM2 -> Hunt pilot number	Passed	
UCJ10Ph2SCUBEG123	Call made from IP phone A to IP phone B via CUBE and multiple transfers made within Unified CM 2	Verify that IP phone A makes call to IP phone B via CUBE through H.323 to H.323 interworking and IP phone B does Multiple transfer with IP phone C and D within Unified CM 2 successfully	Phone A -> Unified CM1 -> H.323 -> CUBE -> H.323 -> Unified CM2 -> Phone B -> Phone C -> Phone D	Passed	
UCJ10Ph2SCUBEG124	Redial by 7841 IP phone via CUBE	Verify that 7841 IP phone makes call to IP phone B via CUBE using Redial option	7841 -> Unified CM1 -> H.323 -> CUBE -> H.323 -> Unified CM2 -> Phone B	Passed	

Cisco Unified Survivable Remote Site Telephony

Logical ID	Title	Description	Call Component Flow	Status	Defects
UCJ10Ph2SSRSTG104	Make call to hunt group sequential in Unified SRST	Verify that hunt group sequential works successfully when calls made within unified SRST	Phone A -> Unified SRST -> Hunt group sequential number	Passed	
UCJ10Ph2SSRSTG210	Call made from 7841 IP phone to other IP phone in SRST mode and consult transfer by other IP phone	Verify that 7841 makes call to IP phone B and IP phone B transfers the call to IP phone C successfully	7841 -> Unified SRST -> Phone B -> Phone C	Passed	
UCJ10Ph2SSRSTG213	Transferring call to other IP phone using Redial option	Verify that 7841 IP phone attends the incoming call and 7841 transfers the call to other IP phone by pressing "Redial" option in 7841 IP phone	Phone A-> Unified SRST -> 7841 -> Phone B	Passed	

Logical ID	Title	Description	Call Component Flow	Status	Defects
UCJ10Ph2S.SRSTG.215	Call forward in 7841 IP phone and make calls from two IP phones and check call forward in 7841 IP Phone	Verify that call forward set successfully in 7841 IP phone and initially make call from IP phone A to 7841 IP phone and check call gets forwarded successfully and then make call from IP phone B to 7841 and check call forward in 7841 IP Phone	Phone A-> Unified SRST -> 7841 ; Phone B -> Unified SRST -> 7841	Passed	
UCJ10Ph2S.SRSTG.216	Call transfer using IP phones with SIP load in SRST mode	Verify that call gets transferred successfully in IP phone using SIP load	Phone A-> Unified SRST -> Phone B -> Phone C	Passed	

Cisco Unity Connection

Logical ID	Title	Description	Call Component Flow	Status	Defects
UCJ10Ph2S.CUCG.001	E-Mail Quota Notification mail for the user when they exceeds the quota limit	To check the user's mail box in corporate e-mail whether the user received quota notification mail.	Unified IP Phone A -> Unified CM -> Unified IP Phone B -> Unity Connection -> Exchange Server	Passed	

Logical ID	Title	Description	Call Component Flow	Status	Defects
UCJ10Ph2S.CUCG.007	Set the customized quota notification mail	To check the user's mail box in corporate mailbox whether the user received customized quota notification mail	Unified IP Phone A -> Unified CM -> Unified IP Phone B -> Unity Connection -> Exchange Server	Passed	
UCJ10Ph2S.CUCG.008	Set the quota notification task to run as every 10 minutes and check for quota notification mail	To check the user's corporate mailbox whether the user received quota notification mail every 10 minutes	Unity Connection -> Exchange Server	Passed	
UCJ10Ph2S.CUCG.011	Schedule a quota notification task at 00:00 time and check timestamp of quota notification mail	To check the timestamp in quota notification mail in user's corporate mailbox	Unity Connection -> Exchange Server	Passed	
UCJ10Ph2S.CUCG.012	Set the Customized subject in quota notification mail	To check the user's corporate mailbox for quota notification mail with customized subject	Unity Connection -> Exchange Server	Passed	
UCJ10Ph2S.CUCG.017	Enable voicemail as attachments in HTML Intelligent Notification at global level	To verify the user is able to see voicemail as an attachment in HTML Intelligent notification	Unified IP Phone A -> Unified CM -> Unified IP Phone B -> Unity Connection -> Exchange Server	Passed	

Logical ID	Title	Description	Call Component Flow	Status	Defects
UCJ10Ph2S.CUCG.018	Allow voicemail as attachments in HTML Intelligent Notification at specific user level	To verify the user is able to see voicemail as an attachment HTML notification	Unified IP Phone A -> Unified CM -> Unified IP Phone B -> Unity Connection -> Exchange Server	Passed	
UCJ10Ph2S.CUCG.022	Enable announce message status when replying to a message through TUI	To enable announce message status when replying to a message through TUI	Unified IP Phone A -> Unified CM ->Unified IP Phone B -> Unity Connection	Passed	
UCJ10Ph2S.CUCG.024	Disable announce message playing status to users while replying through TUI	To disable announce message status when replying to a message through TUI	Unified IP Phone A -> Unified CM ->Unified IP Phone B -> Unity Connection	Passed	
UCJ10Ph2S.CUCG.027	Enable the TUI PIN Reset Behavior and user asked to enter the time interval during conditional warning prompt will be played	To retrieve a voicemail through TUI after the user received voicemail from user in a same site and hears password expiration warning prompt	Unified IP Phone A -> Unified CM ->Unified IP Phone B -> Unity Connection	Passed	
UCJ10Ph2S.CUCG.014	Check whether quota notification mail sent with High importance	To check the task execution results and check the number of notification mail sent to user who exceeds the configured quota limit	Unified IP Phone A-> Unified CM-> Unified IP Phone B-> Unity Connection -> Exchange Server	Passed	

Logical ID	Title	Description	Call Component Flow	Status	Defects
UCJ10Ph2S.CUC.G.015	Set Locale specific alert template for quota notification mail	To check whether the user received the quota notification mail in the locale specific template	Unified IP Phone A-> Unified CM-> Unified IP Phone B-> Unity Connection -> Exchange Server	Passed	
UCJ10Ph2S.CUC.G.026	Retrieve a voicemail through TUI after the user received voicemail from user in a same site and hears password expiration warning prompt	To retrieve a voicemail through TUI after the user received voicemail from user in a same site and hears password expiration warning prompt	Unified IP Phone A -> Unified CM ->Unified IP Phone B -> Unity Connection	Passed	
UCJ10Ph2S.CUC.G.028	Disable the TUI PIN behavior and checks for default behavior of warning prompt	To disable the TUI PIN behavior and checks for default behavior of warning prompt	Unified IP Phone A -> Unified CM ->Unified IP Phone B -> Unity Connection	Passed	
UCJ10S.CUC.SR.001	VMPilot number populated in CTI Line Device	Verify that VMPilot number is populated in CTI Line Device	NA	Passed	

Cisco Unified Attendant Console

Logical ID	Title	Description	Call Component Flow	Status	Defects
UCJ10PH2S.CUACG.003	CUAC client tested with Windows 8 OS	Compatibility testing of CUAC client in Windows 8	NA	Passed	

Logical ID	Title	Description	Call Component Flow	Status	Defects
UCJ10PH2SCUACG004	CUAC Server tested with Windows 2012	Compatibility testing of CUAC server in Windows 2012	NA	Passed	
UCJ10PH2SCUACG005	CUAC Server with Windows 2012 is tested with SQL 2012	Compatibility testing of CUAC server in Windows 2012 with SQL 2012 installed in it.	NA	Passed	

Cisco UC Integration™ for Microsoft Lync

Logical ID	Title	Description	Call Component Flow	Status	Defects
UCJ10Ph2SCUCIG001	Make a call from Docked Window after positioning it to top left side or top right side	Make a Call from Cisco UC Integration™ for Microsoft Lync client from Docked Window after positioning it top of the screen.	Cisco UC Integration™ for Microsoft Lync 1->Unified CM ->Unified IP Phone	Passed	
UCJ10Ph2SCUCIG005	Check missed call indicator in Cisco UC Integration™ for Microsoft Lync docked window	To verify whether the user is able to see the missed call indicator in Cisco UC Integration™ for Microsoft Lync docked window	Unified IP Phone->Unified CM ->Cisco UC Integration™ for Microsoft Lync	Passed	

Logical ID	Title	Description	Call Component Flow	Status	Defects
UCJ10Ph2SCUCIG007	Check the call history tab in Hub Window of Cisco UC Integration™ for Microsoft Lync	To check the call history tab in hub Window of Cisco UC Integration™ for Microsoft Lync for received, missed, and dialed calls	Cisco UC Integration™ for Microsoft Lync 1->Unified CM ->Cisco UC Integration™ for Microsoft Lync 2	Passed	
UCJ10Ph2SCUCIG008	Check the Microsoft Lync presence status when Cisco UC Integration™ for Microsoft Lync is on call	To check the Microsoft Lync presence status when Cisco UC Integration™ for Microsoft Lync currently active in a call	Cisco UC Integration™ for Microsoft Lync 1->Unified CM ->Cisco UC Integration™ for Microsoft Lync 2 -> Lync Server	Passed	
UCJ10Ph2SCUCIG012	Make a call to Cisco UC Integration™ for Microsoft Lync when softphone and desk phone has different DN	To verify whether the user is able to make a call to both desk phone and softphone (Cisco UC Integration™ for Microsoft Lync)	Unified IP Phone 1->Unified CM ->Cisco UC Integration™ for Microsoft Lync	Passed	
UCJ10Ph2SCUCIG013	Check the date and time details in call history after making the call to another site which routed via Session Initiation Protocol Trunk	To verify whether the user is able to make a call to another site and verify the call history details such as date and time.	Unified IP Phone ->Unified CM 1 -> SIP Trunk -> Unified CM 2 -> Cisco UC Integration™ for Microsoft Lync	Passed	

Logical ID	Title	Description	Call Component Flow	Status	Defects
UCJ10Ph2S.CUCIG.002	Make an Inter-cluster video call routed via Cisco Unified Border Element from Docked Window after positioning it	Make an Inter-cluster call to Cisco UC Integration™ for Microsoft Lync client from Docked Window after positioning it top of the screen.	Cisco UC Integration™ for Microsoft Lync 1->Unified CM 1->SIP trunk ->CUBE->SIP trunk -> Unified CM 2 -> Cisco UC Integration™ for Microsoft Lync 2	Passed	
UCJ10Ph2S.CUCIG.011	Make a call to Cisco UC Integration™ for Microsoft Lync when another Unified IP Phone shared the same DN	To verify whether the user is able to make a call to Cisco UC Integration™ for Microsoft Lync when it shares the same DN.	Unified IP Phone 1->Unified CM -> Cisco UC Integration™ for Microsoft Lync 1 -> Unified IP Phone 2	Passed	

Cisco Unified Communications Manager Express

Logical ID	Title	Description	Call Component Flow	Status	Defects
UCJ10Ph2S.CME.G.125	Transferring the call to IP phone in Unified CM from the IP phone present in Hunt group in Unified CME	Verify that IP phone B present in hunt group transfers the call to IP phone E in Unified CM via SIP trunk	Phone A -> Unified CM -> hunt pilot number; Phone B -> Unified CME -> SIP trunk -> Unified CM -> Phone E	Passed	

Logical ID	Title	Description	Call Component Flow	Status	Defects
UCJ10Ph2S.CME.G.121	Create different pickup-groups in CME and check by making calls within Unified CME	Verify that different pickup-groups is created in CME and the call gets established successfully when call made within Unified CME	Phone A-> Unified CME -> Phone B	Passed	
UCJ10Ph2S.CME.G.124	Hunt group sequential in Unified CME	Verify that hunt group sequential works successfully when calls made within CME	Phone A -> Unified CME -> Hunt pilot number	Passed	
UCJ10Ph2S.CME.G.153	Check call park in Unified CME	Verify IP phone B parks the incoming call and IP phone C retrieves the parked call successfully	Phone A -> Unified CME -> Phone B -> Phone C	Passed	
UCJ10Ph2S.CME.G.157	Call made from 8961 IP phone to IP phone B (SCCP load) in Unified CME	Verify that call is made successfully from 8961 IP phone to IP phone B in Unified CME	8961 -> Unified CME -> Phone A	Passed	
UCJ10Ph2S.CME.G.158	DND in 8961 IP phone in Unified CME	Verify that Do Not Disturb works successfully in 8961 IP phone in Unified CME	Phone A -> Unified CME -> 8961	Passed	

Logical ID	Title	Description	Call Component Flow	Status	Defects
UCJ10Ph2S.CME.G.159	Call park in 8961 IP phone in Unified CME and check call gets parked in IP phone with SIP load	Verify that call gets parked successfully in IP phone 8961 and check IP phone B retrieves the parked call	Phone A -> Unified CME -> 8961 -> Phone B	Passed	

Cisco TelePresence Multipoint Control Unit

Logical ID	Title	Description	Call Component Flow	Status	Defects
UCJ10Ph2S.MCU.G.001	Transfer a guest participant	Verify whether the guest participant(Cisco TelePresence System EX60) in a conference managed by Cisco TelePresence MCU can be transferred to another endpoint Cisco TelePresence SX20 Quick Set	EX90-> Unified CM-> SIP Trunk -> MCU 4510 EX60->Unified CM-> SIP Trunk -> MCU 4510 EX60-> transfer-> Unified CM->SX20 Quick Set	Passed	
UCJ10Ph2S.MCU.G.002	Move a guest participant from one conference to another using MCU 4510 Web-GUI	Verify whether the guest participant can be moved from one conference to another in Cisco TelePresence MCU 4510	EX90 -> Cisco VCS ->SIP Trunk-> Unified CM->SIP Trunk -> MCU 4510 (Conference 1) EX60->Cisco VCS ->SIP Trunk->Unified CM-> SIP Trunk-> MCU 4510 (Conference 2) EX60-> Move from Conference 1 to Conference 2	Passed	

Logical ID	Title	Description	Call Component Flow	Status	Defects
UCJ10Ph2SMCUG003	Retrieve a parked MCU 4510 conference call	Verify whether the Cisco TelePresence System EX60 registered in Cisco Unified Communications Manager can retrieve a parked Cisco TelePresence MCU 4510 conference call	EX90 -> Unified CM -> SIP Trunk -> MCU 4510 8945->Unified CM-> SIP Trunk-> MCU 4510 8945-> park-> Unified CM-> EX60-> retrieve	Passed	
UCJ10Ph2SMCUG004	Audio call after video port usage gets exhausted	Verify whether the Cisco TelePresence SX20 Quick Set registered in Cisco Unified Communications Manager is added as an audio call in Cisco TelePresence MCU 4510 after the video port usage gets exhausted	EX90 ->Unified CM -> SIP Trunk -> MCU 4510 EX60 ->Unified CM-> SIP Trunk -> MCU 4510 SX20 Quick Set -> Unified CM->SIP Trunk-> MCU 4510-> audio conference	Passed	
UCJ10Ph2SMCUG005	After locking conference EX90 should not be able to join the conference	Verify whether the Cisco TelePresence System EX90 registered in Cisco TelePresence Video Communication Server is unable to join the Cisco TelePresence MCU 4510 conference after the conference is locked	EX90 -> Cisco VCS -> SIP Trunk-> Unified CM->SIP Trunk-> MCU 4510 -> conference not initiated	Passed	

Logical ID	Title	Description	Call Component Flow	Status	Defects
UCJ10Ph2S.MCU.G.006	Change the layout of EX90 in the MCU 4510	Verify whether the layout of Cisco TelePresence System EX90 registered in Cisco Unified Communications Manager can be changed during a conference in Cisco TelePresence MCU 4510 Web GUI	NA	Passed	
UCJ10Ph2S.MCU.G.007	Guest participants should disconnect after chairperson exits the conference	Verify whether the guest participant is disconnected from the conference after the chairperson exits the conference in Cisco TelePresence MCU 4510	EX90 -> Unified CM -> SIP Trunk -> MCU 4510-> chairperson ID EX60 -> Unified CM->SIP Trunk-> MCU 4510 -> guest ID SX20 Quick Set-> Unified CM->SIP Trunk-> MCU 4510->guest ID	Passed	
UCJ10Ph2S.MCU.G.008	Scheduled Conference Using MCU 4510	Verify whether the pre-configured SIP endpoints registered in Cisco Unified Communications Manager enter the scheduled conference in Cisco TelePresence MCU 4510	SX20 Quick Set , EX90 , EX60 -> Unified CM->SIP Trunk ->MCU 4510->Scheduled Conference	Passed	

Logical ID	Title	Description	Call Component Flow	Status	Defects
UCJ10Ph2SMCUG009	Scheduled Conference with H323 endpoints Using MCU 4510	Verify whether the pre-configured H323 endpoints registered in Cisco TelePresence Video Communication Server enter the scheduled conference in Cisco TelePresence MCU 4510	SX20 Quick Set , EX90 , EX60(H323) -> Cisco VCS-> MCU 4510->Scheduled Conference	Passed	
UCJ10Ph2SMCUG010	Add an endpoint to the conference using SIP Registrar	Verify whether an endpoint can be added to a conference using SIP registrar in Cisco TelePresence MCU 4510	SX20 Quick Set ->Cisco VCS->MCU 4510	Passed	
UCJ10Ph2SMCUG011	Set the conference visibility to private	Verify whether an conference visibility can be changed to private in Cisco TelePresence MCU 4510	NA	Passed	
UCJ10Ph2SMCUG012	Conference call for H323 endpoints using SIP registrar	Verify whether an conference can be initiated by H323 endpoint(Cisco TelePresence System EX90) registered in Cisco TelePresence Video Communication Server in Cisco TelePresence MCU 4510	SX20 Quick Set (H323) ->Cisco VCS-> MCU 4510	Passed	

Logical ID	Title	Description	Call Component Flow	Status	Defects
UCJ10Ph2S.MCU.G013	Conference call to a EX90 endpoint which has set call forward all	Verify whether the conference call using Cisco TelePresence MCU 4510 can be initiated to Cisco TelePresence System EX90 which has set call forward all to another Cisco TelePresence System EX60	MCU 4510->SIP Trunk->Unified CM-> EX90-> call forward all-> Unified CM->EX60	Passed	
UCJ10Ph2S.MCU.G014	Conference call to a EX90 endpoint which has set call forward busy to Cisco VCS registered EX60	Verify whether the conference call using Cisco TelePresence MCU 4510 can be initiated to Cisco TelePresence System EX90 which has set call forward busy to another Cisco TelePresence System EX60 registered in Cisco TelePresence Video Communication Server	8945-> Unified CM -> EX90 MCU 4510 -> SIP Trunk->Unified CM-> EX90-> call forward busy->Unified CM-> SIP Trunk->Cisco VCS-> EX60	Passed	

Logical ID	Title	Description	Call Component Flow	Status	Defects
UCJ10Ph2SMCUG015	Conference call to a EX90 endpoint which has set call forward Unregistered	Verify whether the conference call using Cisco TelePresence MCU 4510 can be initiated to Cisco TelePresence System EX90 which has set call forward Unregistered to another Cisco TelePresence System EX60	MCU 4510 -> SIP Trunk-> Unified CM-> EX90->call forward Unregistered-> Unified CM->EX60	Passed	
UCJ10Ph2SMCUG016	Conference call to a EX90 endpoint which has set call forward Unregistered to Cisco VCS registered EX60	Verify whether the conference call using Cisco TelePresence MCU 4510 can be initiated to Cisco TelePresence System EX90 which has set call forward Unregistered to another Cisco TelePresence System EX60 registered in Cisco TelePresence Video Communication Server	MCU 4510-> SIP Trunk-> Unified CM-> EX90-> call forward Unregistered-> Unified CM-> SIP Trunk-> Cisco VCS-> EX60	Passed	

Logical ID	Title	Description	Call Component Flow	Status	Defects
UCJ10Ph2S.MCU.G.017	Long duration conference calls using MCU 4510	Verify whether the Cisco TelePresence SX20 Quick Set, Cisco Telepresence System EX90 and Cisco TelePresence System EX60 registered in Cisco TelePresence Video Communications Server can join the conference call by Cisco TelePresence MCU 4510 and stay in the conference for 1 hour	SX20 Quick Set -> EX90 -> EX60-> Cisco VCS ->SIP Trunk->Unified CM->SIP Trunk-> MCU 4510	Passed	
UCJ10Ph2S.MCU.G.018	Long duration conference calls using MCU 4510 SIP registrar	Verify whether the Cisco TelePresence SX20 Quick Set, Cisco Telepresence System EX90 and Cisco TelePresence System EX60 registered in Cisco TelePresence Video Communications Server can join the conference call by Cisco TelePresence MCU 4510 by using SIP registrar for long duration (1 hour)	SX20 Quick Set -> EX90-> EX60-> Cisco VCS-> MCU 4510	Passed	

Logical ID	Title	Description	Call Component Flow	Status	Defects
UCJ10Ph2S.MCU.G.019	Conference call to a EX90 endpoint which has set call forward busy	Verify whether the conference call using Cisco TelePresence MCU 4510 can be initiated to Cisco TelePresence System EX90 which has set call forward busy to another Cisco TelePresence System EX60	8945 -> Unified CM-> EX90 MCU 4510 ->SIP Trunk -> Unified CM-> EX90-> call forward busy ->Unified CM-> EX60	Passed	

Cisco TelePresence Management Suite

Logical ID	Title	Description	Call Component Flow	Status	Defects
UCJ10Ph2S.TMS.G.001	Initiate a scheduled conference using Cisco TMS for Unified CM registered endpoints	Verify whether the scheduled conference for Cisco TelePresence System EX90 and Cisco TelePresence System EX60 registered in Cisco Unified Communications Manager can be scheduled successfully using Cisco TelePresence Management Suite	Cisco TMS -> Cisco MCU-> SIP Trunk -> Unified CM-> EX90-> EX60	Passed	

Logical ID	Title	Description	Call Component Flow	Status	Defects
UCJ10Ph2S.TMS.G.002	Initiate a scheduled conference using Cisco TMS for Cisco VCS registered endpoints	Verify whether the scheduled conference for Cisco TelePresence System EX90 and Cisco TelePresence System EX60 registered in Cisco Video Communication Server can be scheduled successfully using Cisco TelePresence Management Suite	Cisco TMS -> Cisco MCU->Cisco VCS-> EX90-> EX60	Passed	
UCJ10Ph2S.TMS.G.003	Mute on join video and audio for a scheduled conference using Cisco TMS for Unified CM registered endpoints	Verify whether mute on join for video and audio for scheduled conference for Cisco TelePresence System EX90 and Cisco TelePresence System EX60 registered in Cisco Unified Communications Manager can be scheduled successfully using Cisco TelePresence Management Suite	Cisco TMS ->Mute on Join->Cisco MCU-> Unified CM-> EX90 -> EX60	Passed	

Logical ID	Title	Description	Call Component Flow	Status	Defects
UCJ10Ph2S.TMS.G.004	Mute on join video and audio for a scheduled conference using Cisco TMS for Cisco VCS registered endpoints	Verify whether mute on join for video and audio for scheduled conference for Cisco TelePresence System EX90 and Cisco TelePresence System EX60 registered in registered in Cisco Video Communication Server can be scheduled successfully using Cisco TelePresence Management Suite	Cisco TMS ->Mute on Join->Cisco MCU-> Cisco VCS->EX90-> EX60	Passed	
UCJ10Ph2S.TMS.G.005	Initiate a scheduled audio conference using Cisco TMS for Unified CM registered endpoints	Verify whether the scheduled audio conference for Cisco TelePresence System EX90 and Cisco TelePresence System EX60 registered in Cisco Unified Communications Manager can be scheduled successfully using Cisco TelePresence Management Suite	Cisco TMS ->Audio->Cisco MCU->Unified CM->EX90-> EX60	Passed	

Logical ID	Title	Description	Call Component Flow	Status	Defects
UCJ10Ph2S.TMS.G.006	Initiate a scheduled audio conference using Cisco TMS for Cisco VCS registered endpoints	Verify whether the scheduled audio conference for Cisco TelePresence System EX90 and Cisco TelePresence System EX60 registered in Cisco Video Communication Server can be scheduled successfully using Cisco TelePresence Management Suite	Cisco TMS ->Audio->Cisco MCU->Cisco VCS->EX90-> EX60	Passed	
UCJ10Ph2S.TMS.G.007	Send message to conference participants during scheduled conference	Verify whether message can be sent to all the participants in the scheduled conference using Conference Control Center in Cisco TelePresence management Suite	Cisco TMS ->Cisco MCU->Cisco VCS->EX90-> EX60	Passed	

Logical ID	Title	Description	Call Component Flow	Status	Defects
UCJ10Ph2S.TMS.G.008	Add a participant to the conference using Conference Control Center	Verify whether the participant can be added in the scheduled conference using Conference Control Center in Cisco TelePresence management Suite	Cisco TMS ->Cisco MCU->Cisco VCS->EX90	Passed	

Cisco TelePresence Conductor

Logical ID	Title	Description	Call Component Flow	Status	Defects
UCJ10Ph2S.TPC.G.001	Video conference between Cisco VCS registered Video end points using multiway with Conductor	Verify the user can able to initiate the video conference between Cisco TelePresence Video Communication Server registered Video end points using multiway with Cisco Telepresence conductor	EX60 -> EX90 -> SX20 Quick Set-> Cisco VCS -> TPC -> MCU 5310	Passed	

Logical ID	Title	Description	Call Component Flow	Status	Defects
UCJ10Ph2S.TPC.G.002	Forward all from EM enabled EX90 to rendezvous conference	Verify the user can able to forward the call from Extension Mobility enabled Cisco TelePresence System EX90 to rendezvous conference	9971 -> Unified CM -> EX90 (EM)-> CFA -> Unified CM -> SIP Trunk -> Telepresence conductor -> Conference	Passed	
UCJ10Ph2S.TPC.G.003	Forward all from Unified CM registered EX60 to rendezvous conference	Verify the user can able to forward the call from Cisco Unified Communications Manager registered Cisco TelePresence System EX60 to rendezvous conference	9971 -> Unified CM -> EX60 -> CFA -> Unified CM -> SIP Trunk -> Telepresence conductor -> Conference	Passed	
UCJ10Ph2S.TPC.G.004	Initiate the long duration conference between Unified CM registered video end points using Telepresence Server via Conductor	Verify the user can able to initiate the long duration conference between Cisco Unified Communications Manager registered video end points using virtual Cisco Telepresence Server via Cisco Telepresence conductor	EX60 -> EX90 -> SX20 Quick Set -> Unified CM -> SIP Trunk -> Telepresence conductor -> Telepresence Server -> conference	Passed	

Logical ID	Title	Description	Call Component Flow	Status	Defects
UCJ10Ph2S.TPC.G.005	Initiate the rendezvous conference from shared line C series endpoints	Verify the user can able to initiate the rendezvous conference from Cisco Unified Communications Manager registered shared line C series end points and Cisco Unified IP Phone 9971	Quick Set C20 and Integrator Package C90 (shared line) ; Quick Set C20 -> 9971 -> Unified CM -> SIP Trunk -> Telepresence conductor ->Conference	Passed	
UCJ10Ph2S.TPC.G.006	Initiate the rendezvous conference from shared line Quick Set C20 then barge the same conference call on Integrator Package C90	Verify the user can able to initiate the rendezvous conference from Cisco Unified Communications Manager registered shared line Cisco TelePresence System Quick Set C20 then barge the same conference call on Cisco TelePresence System Integrator Package C90	Quick Set C20 and Integrator Package C90 (shared line) ; Quick Set C20 -> EX90 -> Unified CM -> SIP Trunk -> Telepresence conductor -> rendezvous conference -> Integrator Package C90 -> Call barge	Passed	

Logical ID	Title	Description	Call Component Flow	Status	Defects
UCJ10Ph2S.TPC.G.007	Lecture type long duration video conference between Cisco VCS registered video end points using Telepresence Server via Telepresence conductor	Verify the user can initiate and maintain the lecture type Long duration video conference between Cisco TelePresence Video Communication Server registered video end points using Cisco Telepresence Server via Cisco Telepresence conductor	EX90 -> Quick Set C20 ->Integrator Package C90 -> Cisco VCS -> Telepresence conductor -> Cisco TelePresence Server -> lecture type conference	Passed	
UCJ10Ph2S.TPC.G.008	Long duration audio conference between Unified CM registered Unified IP Phones using Telepresence Server via Telepresence conductor	Verify the user can initiate and maintain the long duration audio conference between Cisco Unified Communications Manager registered Cisco Unified IP Phone using Cisco Telepresence Server via Cisco Telepresence conductor	6941-> 7961 -> 6921 -> Unified CM -> SIP Trunk -> Telepresence conductor -> Cisco Telepresence Server -> audio conference	Passed	

Logical ID	Title	Description	Call Component Flow	Status	Defects
UCJ10Ph2S.TPC.G.009	Long duration conference between Cisco VCS registered video end points using Telepresence Server via Telepresence conductor	Verify the user can able to maintain the long duration conference between Cisco TelePresence Video Communication Server registered video end points using Cisco Telepresence Server via Cisco Telepresence conductor	EX90 ->Integrator Package C90 -> Quick Set C20 -> Cisco VCS -> Telepresence conductor -> Telepresence Server -> conference	Passed	
UCJ10Ph2S.TPC.G.010	Long duration conference with modified bandwidth between Cisco VCS registered video end points using Cisco TS 7010 via Telepresence conductor	Verify the user can initiate the long duration conference with modified bandwidth between Cisco TelePresence Video Communication Server registered video end points using Cisco Telepresence Server 7010 via Cisco Telepresence conductor	SX20 Quick Set -> EX60 -> EX90 -> Cisco VCS -> Telepresence conductor -> Cisco TS 7010 -> conference	Passed	

Logical ID	Title	Description	Call Component Flow	Status	Defects
UCJ10Ph2S.TPC.G.011	Long duration rendezvous conference between Unified CM registered EM enabled Video End points	Verify the user can able to initiate a long duration rendezvous video conference between Cisco Unified Communications Manager registered Extension Mobility enabled Integrator Package C90 and Quick Set C20	Integrator Package C90 -> Unified CM -> Integrator Package C90 (EM Login) -> Quick Set C20 -> Unified CM -> SIP Trunk -> Telepresence conductor -> Media resource -> rendezvous conference	Passed	
UCJ10Ph2S.TPC.G.012	Make a rendezvous conference between EM enabled Video End points registered in Unified CM	Verify the user can able to make a rendezvous video conference between Cisco Unified Communications Manager registered Extension Mobility enabled Integrator Package C90 and Quick Set C20	Integrator Package C90 -> Unified CM -> Integrator Package C90 (EM Login) -> Quick Set C20 -> Unified CM -> SIP Trunk -> Telepresence conductor -> Media resource -> rendezvous conference	Passed	

Logical ID	Title	Description	Call Component Flow	Status	Defects
UCJ10Ph2S.TPC.G.013	Make a conference between EM enabled Video End points registered in Unified CM	Verify the user can able to make an adhoc conference between Cisco Unified Communications Manager registered Extension Mobility enabled Cisco Unified IP Phone 8941	8941 -> Unified CM -> 8941(EM Login) -> EX60 -> 9971 -> Unified CM -> SIP Trunk -> Telepresence conductor -> Media resource -> rendezvous conference	Passed	
UCJ10Ph2S.TPC.G.014	Park the rendezvous conference call from 8941 and retrieve it on shared line Quick Set C20	Verify the user can able to park the rendezvous conference call from Cisco Unified Communications Manager registered Cisco Unified IP Phone 8941 and then retrieve it on shared line Cisco TelePresence System Quick Set C20	Quick Set C20 -> Unified CM -> Integrator Package C90 -> shared line ; 8941 -> 8945 -> Unified CM -> SIP Trunk -> Telepresence conductor ->media resource -> rendezvous conference ; 8941 -> Park->Unified CM -> Quick Set C20 -> Retrieve -> Integrator Package C90 ->call barge	Passed	

Logical ID	Title	Description	Call Component Flow	Status	Defects
UCJ10Ph2S.TPC.G.015	Park the rendezvous conference call from 8941 and retrieve it on shared line Quick Set C20, then share the presentation	Verify the user can able to park the rendezvous conference call from Cisco Unified Communications Manager registered Cisco Unified IP Phone 8941 and then retrieve it on shared line Cisco TelePresence System Quick Set C20, then share the presentation	Quick Set C20 -> Integrator Package C90 -> shared line; 8941 -> 8945 -> Unified CM -> SIP Trunk -> Telepresence conductor -> media resource -> rendezvous conference; 8945 -> Park->Unified CM -> Quick Set C20 -> Retrieve -> Integrator Package C90 ->call barge -> presentation sharing	Passed	
UCJ10Ph2S.TPC.G.016	Park the rendezvous conference call from 9971 and retrieve it on Integrator Package C90 then share the presentation	Verify the user can able park the rendezvous conference call from Cisco Unified Communications Manager registered Cisco Unified IP Phone 9971 and then retrieve it on Cisco TelePresence System Integrator Package C90 then share the presentation	8941 -> 9971 ->Unified CM -> SIP Trunk -> TelePresence conductor -> media resource -> rendezvous conference; 9971 -> Park->Unified CM -> Quick Set C20 -> Retrieve -> sharing the presentation	Passed	

Logical ID	Title	Description	Call Component Flow	Status	Defects
UCJ10Ph2S.TPC.G.017	Park the rendezvous conference call from DX650 and retrieve it on EM enabled Quick Set C20, then sharing the presentation	Verify the user can able to park the rendezvous conference call from Cisco Unified Communications Manager registered DX650 and then retrieve it on extension mobility enabled Cisco TelePresence System Quick Set C20 then sharing the presentation	Quick Set C20 -> Unified CM -> 8941(EM Login) -> DX650 -> Unified CM -> SIP Trunk -> Telepresence conductor -> media resource -> rendezvous conference; DX650->Park -> Unified CM -> Quick Set C20 -> Retrieve -> presentation sharing	Passed	
UCJ10Ph2S.TPC.G.018	Switch to secondary Telepresence conductor when primary conductor goes down in long duration rendezvous video conference between Cisco VCS registered end points	Verify the secondary Telepresence conductor can support in the long duration rendezvous video conference between Cisco TelePresence Video Communication Server registered end points	EX60 -> Quick Set C20 ->SX20 Quick Set -> Cisco VCS-> TPC-> MCU	Passed	

Logical ID	Title	Description	Call Component Flow	Status	Defects
UCJ10Ph2S.TPC.G.019	Switch to secondary Telepresence conductor when primary conductor goes down in long duration rendezvous video conference between Unified CM registered IP Phones	Verify the secondary Telepresence conductor can support for the long duration rendezvous video conference between Cisco Unified Communications Manager registered Cisco Unified IP Phones	8941-> 8945-> 9971-> Unified CM -> TPC -> MCU	Passed	
UCJ10S.TPC.G.037	Transfer the Conference call from Unified CM registered EX/C series phones to 8941 IP Phone	Verify the user can transfer the Conference call from Cisco Unified Communications Manager registered EX/C series phones to Cisco Unified IP Phone 8941	EX/C Series -> Unified CM -> SIP Trunk -> TP Conductor -> Media Resource -> rendezvous Conference -> EX/C Series -> Transfer ->Unified CM -> 8941 -> Conference	Failed	CSCul04054

Cisco TelePresence Server

Logical ID	Title	Description	Call Component Flow	Status	Defects
UCJ10Ph2S.TS.G.001	Initiate the Meet Me video conference from Quickset C20 registered with Cisco VCS using Cisco TS 7010 registered with Cisco VCS as a SIP registrar	To verify that user can able to initiate the Meet Me video conference from Quickset C20 registered with Cisco VCS using Cisco TelePresence Server 7010 registered with Cisco VCS as a SIP registrar	Quickset C20 -> Cisco VCS-> Cisco TS 7010	Passed	
UCJ10Ph2S.TS.G.002	Initiate the Meet Me video conference from Integrator Package C90 registered with Unified CM using Cisco TS 7010 registered with Cisco VCS as a SIP registrar	To verify that user can able to initiate the Meet Me video conference from Integrator Package C90 registered with Unified CM using Cisco TelePresence Server 7010 registered with Cisco VCS as a SIP registrar	Integrator Package C90 -> Unified CM-> SIP Trunk -> Cisco VCS-> Cisco TS 7010	Passed	

Logical ID	Title	Description	Call Component Flow	Status	Defects
UCJ10Ph2S.TS.G.003	Initiate the Meet Me video conference from IP Phone 8945 registered with Unified CM using Cisco TS 7010 registered with Cisco VCS as a SIP registrar	To verify that user can able to initiate the Meet Me video conference from IP Phone 8945 registered with Unified CM using Cisco TelePresence Server 7010 registered with Cisco VCS as a SIP registrar	8945 -> Unified CM -> SIP Trunk-> Cisco VCS->Cisco TS 7010	Passed	
UCJ10Ph2S.TS.G.004	Initiate the Meet Me audio conference from SX20 Quickset registered with Cisco VCS using Cisco TS 7010 registered with Cisco VCS as a SIP registrar	To verify that user can able to initiate the Meet Me audio conference from SX20 Quickset registered with Cisco VCS using Cisco TelePresence Server 7010 registered with Cisco VCS as a SIP registrar	SX20 Quickset (set audio) -> Cisco VCS -> Cisco TS 7010	Passed	
UCJ10Ph2S.TS.G.005	Initiate the Meet Me audio conference from 6961 registered with Unified CM using Cisco TS 7010 registered with Cisco VCS as a SIP registrar	To verify that user can able to initiate the Meet Me audio conference from 6961 registered with Unified CM using Cisco TelePresence Server 7010 registered with Cisco VCS as a SIP registrar	6961 -> Unified CM-> SIP Trunk -> Cisco VCS-> Cisco TS 7010	Passed	

Logical ID	Title	Description	Call Component Flow	Status	Defects
UCJ10Ph2S.TS.G.006	Initiate the Meet Me conference from Quickset C20 registered with Unified CM using EM enabled user via Cisco TS 7010	To verify that user can able to initiate the Meet Me conference from Quickset C20 registered with Unified Cm using EM enabled user via Cisco TelePresence Server 7010	Quickset C20 -> Unified CM -> SIP Trunk -> Cisco TS 7010	Passed	
UCJ10Ph2S.TS.G.007	Make an adhoc conference between Unified CM registered EM enabled Video Endpoints	Verify the user can able make an adhoc video conference between Cisco Unified Communications Manager registered Extension Mobility enabled Video Endpoints	Quickset C20, EX60 and EX90(EM) -> Unified CM-> Conductor->SIP Trunk-> Cisco TS 7010	Passed	
UCJ10Ph2S.TS.G.008	Transfer the Meet Me video conference call from Unified CM registered EM enabled Quickset C20 to 8945	Verify the user can able transfer the Meet Me video conference call from Cisco Unified Communications Manager registered EM enabled Quickset C20 to Cisco Unified IP Phone 8945	Quickset C20->Unified CM->SIP Trunk -> Cisco TS 7010-> Initiate video conference Quickset C20-> Transfer-> 8945	Passed	

Logical ID	Title	Description	Call Component Flow	Status	Defects
UCJ10Ph2S.TS.G.009	Transfer the Meet Me video conference call from Unified CM registered 8941 to EM enabled EX60	Verify the user can able transfer the rendezvous conference call from Cisco Unified Communications Manager registered Cisco Unified IP Phone 8941 to EM enabled EX60	8941->Unified CM -> SIP Trunk-> Cisco TS 7010->Initiate conference 8941->Transfer -> EX60	Passed	
UCJ10Ph2S.TS.G.010	Meet Me video conference between video endpoints registered with Cisco VCS using Cisco TS 7010 registered with Cisco VCS as a SIP registrar	To verify that user can able to do Meet Me video conference between video endpoints registered with Cisco VCS using Cisco TelePresence Server 7010 registered with Cisco VCS as a SIP registrar	Quickset C20 , EX60 and EX90-> Cisco VCS -> Cisco TS 7010	Passed	
UCJ10Ph2S.TS.G.011	Meet Me video conference between IP Phone 8900 Series registered with Unified CM using Cisco TS 7010 registered with Cisco VCS as a SIP registrar	To verify that user can able to do the Meet Me video conference between IP Phone 8900 series registered with Unified CM using Cisco TelePresence Server 7010 registered with Cisco VCS as a SIP registrar	8941 , 8945 and8961-> Unified CM-> SIP Trunk-> Cisco VCS-> Cisco TS 7010	Passed	

Logical ID	Title	Description	Call Component Flow	Status	Defects
UCJ10Ph2S.TS.G.012	Meet Me video conference between video endpoints registered with Unified CM using Cisco TS 7010 registered with Cisco VCS as a SIP registrar	To verify that user can able to do Meet Me video conference between video endpoints registered with Unified CM using Cisco TelePresence Server 7010 registered with Cisco VCS as a SIP registrar	Quickset C20 , EX60 and EX90-> Unified CM-> SIP Trunk->Cisco VCS -> Cisco TS 7010	Passed	
UCJ10Ph2S.TS.G.013	Transfer the Meet Me video conference from Unified CM Registered Quickset C20 to 8945	To verify that user can able to transfer the Meet Me video conference from Unified CM registered Quickset C20 to 8945 using Cisco TelePresence Server 7010 registered with Cisco VCS as a SIP registrar	Quickset C20-> Unified CM-> SIP Trunk-> Cisco VCS -> Cisco TS 7010 ; Quickset C20 -> Transfer->8945	Passed	
UCJ10Ph2S.TS.G.014	Transfer the Meet Me video conference from Cisco VCS registered Quickset C20 to EX60	To verify that user can able to transfer the Meet Me video conference from Unified CM registered Quickset C20 to EX60 using Cisco TelePresence Server 7010 registered with Cisco VCS as a SIP registrar	Quickset C20 -> Cisco VCS -> Cisco TS 7010 Quickset C20 -> Transfer-> EX60	Passed	

Logical ID	Title	Description	Call Component Flow	Status	Defects
UCJ10Ph2S.TS.G.015	Transfer the Meet Me video conference from EX60 registered with Cisco VCS to 9971 registered with Unified CM	To verify that user can able to transfer the Meet Me video conference from Unified CM registered Quickset C20 to 9971 using Cisco TelePresence Server 7010 registered with Cisco VCS as a SIP registrar	EX60 -> Cisco VCS -> Cisco TS 7010 EX60 ->Transfer->9971	Passed	
UCJ10Ph2S.TS.G.016	Transfer the video conference between video endpoints registered with Cisco VCS while sharing the presentation	To verify that user can able to Transfer the video conference between video endpoints registered with Cisco VCS while sharing the presentation	Quickset C20 , EX60 -> Cisco VCS-> Cisco TS 7010 ->Presentation Sharing ; Quickset C20->Transfer->EX90	Passed	
UCJ10Ph2S.TS.G.017	Meet Me video conference between Unified CM registered endpoints and Cisco VCS registered endpoints	To verify that user can able to Initiate the Meet Me video conference between Unified CM Registered endpoints and Cisco VCS registered endpoints using Cisco TelePresence Server 7010 registered with Cisco VCS as a SIP registrar	Quickset C20 , EX60 -> Cisco VCS-> Cisco TS 7010 ; 8945 and 8941-> Unified CM-> SIP Trunk->Cisco VCS -> Cisco TS 7010	Passed	

Logical ID	Title	Description	Call Component Flow	Status	Defects
UCJ10Ph2S.TS.G.018	Meet Me conference from Quickset C20 registered with Unified CM using EM enabled user via Cisco TS 7010 and share the presentation while in video conference	To verify that user can able to initiate the Meet Me conference from Quickset C20 registered with Unified CM using EM enabled user via Cisco TelePresence Server 7010 and share the presentation while in video conference	Quickset C20-> Unified CM -> SIP Trunk -> Cisco TS 7010 -> Presentation Sharing	Passed	
UCJ10Ph2S.TS.G.019	Meet Me video conference from video endpoint registered with Cisco VCS as a H.323 endpoint	To verify that user can able to initiate the Meet Me video conference from Quickset C20 registered with Cisco VCS as a H.323 endpoint using Cisco TelePresence Server 7010	Quickset C20 (H.323) -> Cisco VCS-> Cisco TS 7010	Passed	
UCJ10Ph2S.TS.G.020	Meet Me video conference between video endpoints registered with Cisco VCS as a H.323 endpoint and Share the Presentation	To verify that user can able to initiate the Meet Me video conference between Quickset C20 ,EX60 and EX90 registered with Cisco VCS as a H.323 endpoints and sharing the presentation	Quickset C20, EX60 and EX90 (H.323) -> Cisco VCS-> Cisco TS 7010 -> Presentation Sharing	Passed	

Logical ID	Title	Description	Call Component Flow	Status	Defects
UCJ10Ph2S.TS.G.021	Meet Me conference between Cisco VCS registered Quickset C20 (SIP endpoint) and EX60 (H.323 endpoint)	To verify that user can able to initiate the Meet Me video conference between Cisco VCS registered Quickset C20 as a SIP endpoint and EX60 as a H.323 endpoint using Cisco TS 7010	Quickset C20 (SIP) and EX90 (H.323) -> Cisco VCS->Cisco TS 7010	Passed	
UCJ10Ph2S.TS.G.022	Transfer conference call from Cisco VCS registered Quickset C20 (SIP endpoint) to EX60 (H.323 endpoint)	To verify that user can able to transfer the Meet Me video conference from Cisco VCS registered Quickset C20 as a SIP endpoint to EX60 as a H.323 endpoint using Cisco TS 7010	Quickset C20 (SIP) -> Cisco VCS-> Cisco TS 7010 Quickset C20(SIP) -> Transfer-> EX60 (H.323)	Passed	
UCJ10Ph2S.TS.G.023	Presentation sharing in Meet Me video conference between Cisco VCS registered Quickset C20 (SIP endpoint) and EX60 (H.323 endpoint)	To verify that user can able to Share the Presentation in Meet Me video conference between Cisco VCS registered Quickset C20 as a SIP endpoint and EX60 as a H.323 endpoint using Cisco TS 7010	Quickset C20 (SIP) and EX90 (H.323) ->Cisco VCS-> Cisco TS 7010	Passed	

Logical ID	Title	Description	Call Component Flow	Status	Defects
UCJ10Ph2S.TS.G.024	Park and retrieve the Meet Me video conference from IP Phone 8945 registered with Unified CM	To verify that user can able to park and retrieve the Meet Me video conference from IP Phone 8945 using Cisco TS 7010	8945-ph1-> Unified CM->SIP Trunk ->Cisco TS7010 ; 8945 - ph1-> park -> Unified CM-> 8945 - ph2 -> retrieve	Passed	
UCJ10Ph2S.TS.G.025	Park the Meet Me video conference from IP Phone 8945 registered with Unified CM and Retrieve the conference from Quickset C20	To verify that user can able to park the Meet Me video conference from IP Phone 8945 and retrieve the conference Quickset C20 using Cisco TS 7010	8945 -> Unified CM -> SIP Trunk-> Cisco TS 7010 ; 8945 -> park -> Unified CM -> Quickset C20 -> retrieve	Passed	

Cisco Collaboration Expressway

Logical ID	Title	Description	Call Component Flow	Status	Defects
UCJ10Ph2S.CE.G.001	Video Call between EX60 and EX90 via Collaboration Edge	Verify the user can able to make a video call between Cisco TelePresence System EX60 and Cisco TelePresence System EX90, which is in remote network registered in Cisco Unified Communications Manager via Collaboration Edge	EX60-> EX90 -> Cisco VCS-E -> Cisco VCS-C -> Unified CM -> Video call	Passed	

Logical ID	Title	Description	Call Component Flow	Status	Defects
UCJ10Ph2S.CE.G.002	Hold and Resume the video call between EX60 and EX90 via Collaboration Edge	Verify the user can able to hold and resume a video call between Cisco TelePresence System EX60 and Cisco TelePresence System EX90, which is in remote network registered in Cisco Unified Communications Manager via Collaboration Edge	EX60-> EX90 -> Cisco VCS-E -> Cisco VCS-C -> Unified CM -> Video call	Passed	
UCJ10Ph2S.CE.G.003	Web Snapshots when video call between EX60 and EX90 via Collaboration Edge	Verify the user can able to allow web snapshots when video call between Cisco TelePresence System EX60 and Cisco TelePresence System EX90, which is in remote network registered in Cisco Unified Communications Manager via Collaboration Edge	EX60-> EX90 -> Cisco VCS-E -> Cisco VCS-C -> Unified CM -> Video call	Passed	

Logical ID	Title	Description	Call Component Flow	Status	Defects
UCJ10Ph2S.CE.G.004	Video Call Transfer from EX60 to EX90 via Collaboration Edge	Verify the user can able to make a video call transfer from Cisco TelePresence System EX60 to Cisco TelePresence System EX90, which is in remote network registered in Cisco Unified Communications Manager via Collaboration Edge	EX60-> EX90-1 -> EX90-2 -> Cisco VCS-E -> Cisco VCS-C -> Unified CM -> Video call-> EX60-> EX90-1	Passed	
UCJ10Ph2S.CE.G.005	Video Call from Quick Set C20 (corporate) to EX90 (remote) via Collaboration Edge	Verify the user can able to make a video call from Cisco TelePresence System Quick Set C20 in corporate network, to Cisco TelePresence System EX90 in remote network registered in Cisco Unified Communications Manager via Collaboration Edge	EX90 -> Cisco VCS-E -> Cisco VCS-C -> Unified CM; Quick Set C20 -> Unified CM -> EX90 -> Video call	Passed	

Logical ID	Title	Description	Call Component Flow	Status	Defects
UCJ10Ph2S.CE.G.006	Modified bandwidth video call between EX90 (remote) and Quick Set C20 (corporate) via Collaboration Edge	Verify the user can able to modify video call bandwidth from Cisco TelePresence System Quick Set C20 in corporate network, to Cisco TelePresence System EX90 in remote network registered in Cisco Unified Communications Manager via Collaboration Edge	EX90 -> Cisco VCS-E -> Cisco VCS-C -> Unified CM; Quick Set C20 -> Unified CM -> EX90 -> Modified Bandwidth -> Video call	Passed	
UCJ10Ph2S.CE.G.007	Video Call from shared line EX90 (corporate) to Cisco Telepresence System EX90 (remote) via Collaboration Edge	Verify the user can able to make a video call from shared line Cisco TelePresence System EX90 in corporate network, to Cisco TelePresence System EX90 in remote network registered in Cisco Unified Communications Manager via Collaboration Edge	EX90 -> Cisco VCS-E -> Cisco VCS-C -> Unified CM ; EX60-> EX90 -> Unified CM -> Shared line ; EX90 -> Unified CM -> EX90 -> Video call	Passed	

Logical ID	Title	Description	Call Component Flow	Status	Defects
UCJ10Ph2S.CE.G.008	Call forward all from EX60 (corporate) to EX90 (remote) via Collaboration Edge	Verify the user can able to forward all the video call from Cisco TelePresence System EX60 in corporate network, to Cisco TelePresence System EX90 in remote network registered in Cisco Unified Communications Manager via Collaboration Edge	EX90 -> Cisco VCS-E -> Cisco VCS-C -> Unified CM; Quick Set C20 -> Unified CM -> EX60 -> EX90	Passed	
UCJ10Ph2S.CE.G.009	Consultative Call transfer from SX20 Quick Set (corporate) to EX60 (remote) via Collaboration Edge	Verify the user can able to perfom consultative transfer of video call from Cisco TelePresence SX20 Quick Set in corporate network, to Cisco TelePresence System EX60 in remote network registered in Cisco Unified Communications Manager via Collaboration Edge	EX60 -> Cisco VCS-E -> Cisco VCS-C -> Unified CM; Quick Set C20 -> Unified CM -> SX20 Quick Set -> EX60	Passed	

Logical ID	Title	Description	Call Component Flow	Status	Defects
UCJ10Ph2S.CE.G.010	Video Call conference between EX60, EX90 and SX20 Quick Set via Collaboration Edge	Verify the user can able to make a video call conference between Cisco TelePresence System EX60, Cisco TelePresence System EX90 and Cisco TelePresence SX20 Quick Set, which is in remote network registered in Cisco Unified Communications Manager via Collaboration Edge	EX90 -> EX60-> SX20 Quick Set -> Cisco VCS-E -> Cisco VCS-C	Passed	
UCJ10Ph2S.CE.G.011	Presentation sharing in conference call between EX60, EX90 and SX20 Quick Set via Collaboration Edge	Verify the user can able to share the presentation during video call conference between Cisco TelePresence System EX60, Cisco TelePresence System EX90 and Cisco TelePresence SX20 Quick Set, which is in remote network registered in Cisco Unified Communications Manager via Collaboration Edge	EX90 -> EX60-> SX20 Quick Set -> Cisco VCS-E -> Cisco VCS-C	Passed	

Logical ID	Title	Description	Call Component Flow	Status	Defects
UCJ10Ph2S.CE.G.012	Change the presentation layout while conference call between EX60, EX90 and SX20 Quick Set via Collaboration Edge	Verify the user can able to change presentation layout during video call conference between Cisco TelePresence System EX60, Cisco TelePresence System EX90 and Cisco TelePresence SX20 Quick Set, which is in remote network registered in Cisco Unified Communications Manager via Collaboration Edge	EX90 -> EX60-> SX20 Quick Set -> Cisco VCS-E -> Cisco VCS-C	Passed	
UCJ10Ph2S.CE.G.013	Audio Call between 6945 (corporate) and EX90 (remote) via Collaboration Edge	Verify the user can able to make an audio call from Unified IP Phone 6945 in corporate network, to Cisco TelePresence System EX90 in remote network registered in Cisco Unified Communications Manager via Collaboration Edge	EX90 -> Cisco VCS-E -> Cisco VCS-C -> Unified CM;6945 -> Unified CM -> EX90 -> Audio call	Passed	

Logical ID	Title	Description	Call Component Flow	Status	Defects
UCJ10Ph2S.CE.G.014	Video Call from EX90 (remote) via Collaboration Edge to Cisco VCS registered EX60 (corporate)	Verify the user can able to make a video call from Cisco TelePresence System EX90 in remote network registered in Cisco Unified Communications Manager via Collaboration Edge to Cisco TelePresence System EX60 registered in Cisco TelePresence Video Communication Server	EX90 -> Cisco VCS-E -> Cisco VCS-C -> Unified CM; EX90 -> Unified CM -> SIP Trunk -> Cisco VCS-> EX60 -> Video call	Passed	

Logical ID	Title	Description	Call Component Flow	Status	Defects
UCJ10Ph2S.CE.G.015	Call forward all from EX60 (corporate) to EX90 (remote) via Collaboration Edge, when getting the call from Cisco VCS registered Quick Set C20	Verify the user can able to forward all the video call from Cisco TelePresence System EX60 in corporate network, to Cisco TelePresence System EX90 in remote network registered in same Cisco Unified Communications Manager via Collaboration Edge , when getting the call from Cisco TelePresence Video Communication Server registered Cisco TelePresence System Quick Set C20	EX90 -> Cisco VCS-E -> Cisco VCS-C -> Unified CM; Quick Set C20 -> Cisco VCS-> SIP Trunk -> Unified CM -> EX60 -> EX90	Passed	

Logical ID	Title	Description	Call Component Flow	Status	Defects
UCJ10Ph2S.CE.G.016	Call transfer from 8941 (corporate) to EX60 (remote) via Collaboration Edge	Verify the user can able to transfer the video call from Unified IP Phone 8941 in corporate network, to Cisco TelePresence System EX60 in remote network registered in Cisco Unified Communications Manager via Collaboration Edge	EX60 -> Cisco VCS-E -> Cisco VCS-C -> Unified CM; DX650 -> Unified CM -> 8941-> EX60	Passed	
UCJ10Ph2S.CE.G.017	Call Park in 9971 (corporate) and retrieve from EX60 (remote) via Collaboration Edge	Verify the user can able to park the video call in Unified IP Phone 9971 in corporate network, and retrieve from Cisco TelePresence System EX60 in remote network registered in Cisco Unified Communications Manager via Collaboration Edge	EX60 -> Cisco VCS-E -> Cisco VCS-C -> Unified CM; Integrator Package C90 -> Unified CM -> 9971 -> EX60	Passed	

Logical ID	Title	Description	Call Component Flow	Status	Defects
UCJ10Ph2S.CE.G.018	Inter cluster video call through SIP Trunk between 8941 (corporate) and EX90 (remote) via Collaboration Edge	Verify the user can able to make a video call through SIP Trunk between Cisco Unified IP Phone 8941 in corporate network and Cisco TelePresence System EX90 in remote network registered in Cisco Unified Communications Manager via Collaboration Edge	EX90 -> Cisco VCS-E -> Cisco VCS-C -> Unified CM1; 8941 -> Unified CM2 -> SIP Trunk -> Unified CM1 -> EX90 -> Video call	Passed	

Regression

Regression Execution							
Features tested	**Total test cases**	**Passed**	**Passed %**	**Passed W/X**	**Passed W/X %**	**Failed**	**Failed %**
Auto-call Pickup	5	5	100%	0	0%	0	0%
Call Forward All	14	14	100%	0	0%	0	0%
Call Forward Busy	6	6	100%	0	0%	0	0%
Call Park	13	13	100%	0	0%	0	0%
Call Pickup	19	19	100%	0	0%	0	0%
Call Transfer	3	3	100%	0	0%	0	0%
CBarge	3	3	100%	0	0%	0	0%
Conference	7	7	100%	0	0%	0	0%
Conference List	9	9	100%	0	0%	0	0%

Regression Execution							
Features tested	**Total test cases**	**Passed**	**Passed %**	**Passed W/X**	**Passed W/X %**	**Failed**	**Failed %**
DND	1	1	100%	0	0%	0	0%
Hold Reversion	5	5	100%	0	0%	0	0%
iDivert	1	1	100%	0	0%	0	0%
MakeCall	26	26	100%	0	0%	0	0%
MeetMe	1	1	100%	0	0%	0	0%
OPickup	1	1	100%	0	0%	0	0%
Park Reversion	5	5	100%	0	0%	0	0%
Privacy on Hold	1	1	100%	0	0%	0	0%
Redial	5	5	100%	0	0%	0	0%
Shared Line	3	3	100%	0	0%	0	0%
TOTAL	128	128	100%	0	0%	0	0%

Related Documentation

Cisco Unified CM

http://www.cisco.com/en/US/docs/voice_ip_comm/cucm/rel_notes/10_0_1/delta/CUCM_BK_C206A718_00_cucm-new-and-changed-1001.html

Cisco Unified Communications Self Care Portal User Guide 10.0.0

http://www.cisco.com/en/US/docs/voice_ip_comm/cucm/useroptions/10_0_1/CUCM_BK_U263A6A9_00_uc-self-care-user-guide_1001.html

Cisco Unified IP Phones

Cisco IP Phone 78xx - Release Notes

http://www.cisco.com/c/en/us/td/docs/voice_ip_comm/cuipph/7821_7841_7861/firmware/10_1_1/english/release_notes/PA2D_BK_RFC9D799_00_rn-10_1_1-7821-7841-7861/PA2D_BK_RFC9D799_00_rn-10_1_1-7821-7841-7861_chapter_00.html

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http://www.cisco.com/en/US/docs/voice_ip_comm/cuipph/8831/firmware/9_3_3/english/release_notes_new/CS38_BK_R67E3E94_00_rn-9_3_3-8831.pdf

Cisco Desktop Collaboration Experience DX650

Admin Guide

http://www.cisco.com/en/US/docs/voice_ip_comm/cdce/dx600/admin/10_0_1/english/DX60_BK_C989350E_00_cisco-desktop-collaboration-experience-dx600_chapter_01000.html

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Data Sheet

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Quick start

http://www.cisco.com/en/US/docs/voice_ip_comm/cdce/dx600/qs/10_0_1/english/dx650qs1001en.pdf

User Guide

http://www.cisco.com/en/US/docs/voice_ip_comm/cdce/dx600/user/10_0_1/english/DX60_BK_CAA8EFA0_00_cisco-desktop-collaboration-experience-dx650_chapter_01.html

Useful Link

http://www.cisco.com/en/US/prod/voicesw/ps6788/phones/ps12956/ps12959/dx_650_series_video.html

Cisco Unified CME

Design Guide

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Administration Guide

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http://www.cisco.com/en/US/docs/voice_ip_comm/cusrst/admin/sccp_sip_srst/configuration/guide/SCCP_and_SIP_SRST_Admin_Guide.html

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Documentation Guide

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Cisco Unified Border Element

http://www.cisco.com/en/US/docs/ios/voice/cube/configuration/guide/vb_book/vb_book.html

Cisco Jabber for Windows

Installation Guide

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Getting started Guide

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Configuration Guide

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Release Notes

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Deployment Guide

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Installation Guide

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Cisco TelePresence Server on VM

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Installation Guide

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Cisco TelePresence Management Suite

Administration Guide

http://www.cisco.com/c/dam/en/us/td/docs/telepresence/infrastructure/tms/admin_guide/Cisco-TMS-Admin-Guide-14-3-2.pdf

Installation Guide

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Cisco TelePresence SX20 Quickset

Getting Started

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Cisco TelePresence MCU

Administration Guide

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Cisco Collaboration Expressway

Administration Guide

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