



Test Results for Cisco Unified Communications System Release 8.6(2a) for Japanese

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CHAPTER 1

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 build 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 Japanese

Cisco Unified Communications System Test for Japanese, 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 SE 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 Japanese 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.



Note

The current release focusses on testing the UC components which includes Video, Wireless and Desktop Virtualization. Refer to [Environment Matrix](#) for version details.

In this Cisco Unified Communications System Test release for Japanese the following features are tested.

- Cisco Unified Communications Manager
- Cisco TelePresence Video Communication Server
- Cisco TelePresence Video Communication Server Expressway
- Cisco Video IP Phones
- Cisco Virtualization Experience Client
- Cisco Unified Wireless IP Phones
- Cisco Cius
- Cisco Jabber for MAC
- Cisco Jabber for iPhone
- Cisco Unified Contact Center Express
- Cisco Unified Presence
- Cisco Unity Connection
- Cisco UC Integration for Microsoft LYNC
- Cisco Unified Border Element
- Cisco Unified Personal Communicator
- Cisco Unified Personal Communicator for Mac
- Cisco Unified Communications Manager Express
- Analog Telephone Adapter
- Cisco Unified Survivable Remote Site Telephony
- Cisco Unified IP Phone

- Cisco IP Communicator

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 |
| ATA | Analog Telephone Adapter |
| BAT | Bulk Administrator tool |
| BLF | Busy Lamp Field |
| CAD | Cisco Agent Desktop |
| CAD BE | Cisco Agent Desktop - Browser Edition |
| CAS | Channel Associated Signalling |
| CCD | Call Control Discovery |
| CDA | Cisco Desktop Administrator |
| 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 |
| CIPC | Cisco Unified IP Communicator |
| CFNC | Call Forward No Coverage |
| CFUR | Call Forward Unregistered |
| CLI | Command Line Interface |
| CLID | Caller ID |
| CME | Cisco Unified Communications Manager Express |
| CoW | Clustering over WAN |
| CSD | Cisco Supervisor Desktop |
| CSS | Calling Search Space |
| CSQ | Contact Service Queue |
| CTI | Computer Telephony Interface |
| CU | Cisco Unity |
| CUC | Cisco Unity Connection |
| CUCI-LYNC | Cisco UC Integration for LYNC |
| CUCM | Cisco Unified Communications Manager |
| CUCIMOC | Cisco UC Integration for Microsoft Office Communicator |

| Acronym | Description |
|----------------|---|
| CUPC | Cisco Unified Personal Communicator |
| CUPS | Cisco Unified Presence Server |
| DCR | Device and Credential Repository |
| DHCP | Dynamic Host Configuration Protocol |
| DN | Directory Number |
| DND | Do Not Disturb |
| DO | Delayed Offer |
| DPNSS | Digital Private Network Signaling System |
| DSCP | Differentiated Services Code Point. |
| EMCC | Extension Mobility Cross Cluster |
| EO | Early Offer |
| FXS | Foreign Exchange Station |
| GW | Gateway |
| HR | Historical Reporting |
| HA | High Availability |
| ICT | Inter-cluster trunk |
| IPMA | Cisco IP Manager Assistant |
| IPPA | Cisco Unified IP Phone Agent |
| IPPM | IP Phone Messenger |
| ISDN | Integrated Services Digital Network |
| MGCP | Media Gateway Control Protocol |
| MOH | Music on hold |
| MWI | Message Waiting Indicator |
| NLP | Non Linear Processing |
| NTP | Network Time Protocol |
| POTS | Plain Old Telephony System |
| PCA | Personal Communication Assistant |
| PCoIP | PC over IP |
| PRI | Primary Rate Interface |
| PSTN | Public Switched Telephone Network |
| RSS | Really Simple Syndication |
| QRT | Quality Report Tool |
| QSIG | Q-Signaling protocol |
| SAF | Service Advertisement Framework |
| SIP | Session Initiation Protocol |
| SME | Cisco Unified Communications Manager Session Management Edition |
| SCCP | Skinny Client Control Protocol |

| Acronym | Description |
|----------------|---|
| SRST | Survivable Remote Site Telephony |
| SSL | Security Socket layer |
| TNP | The New Phone |
| TRP | Trust Relay Point |
| TUI | Telephony User Interface |
| UCS | Unified Computing System |
| UCCX | Cisco Unified Contact Center Express |
| UMG | Unified Messaging Gateway |
| VCS | Cisco TelePresence Video Communication Server |
| VGW | Voice Gateway |
| VoIP | Voice over IP |
| VPIM | Voice Profile for Instant Messaging |
| VMN | Voice Mail Notification |
| VXC | Virtualized Experience Client |
| VXI | Virtualized Experience Infrastructure |
| WAN | Wide Area Network |



CHAPTER 2

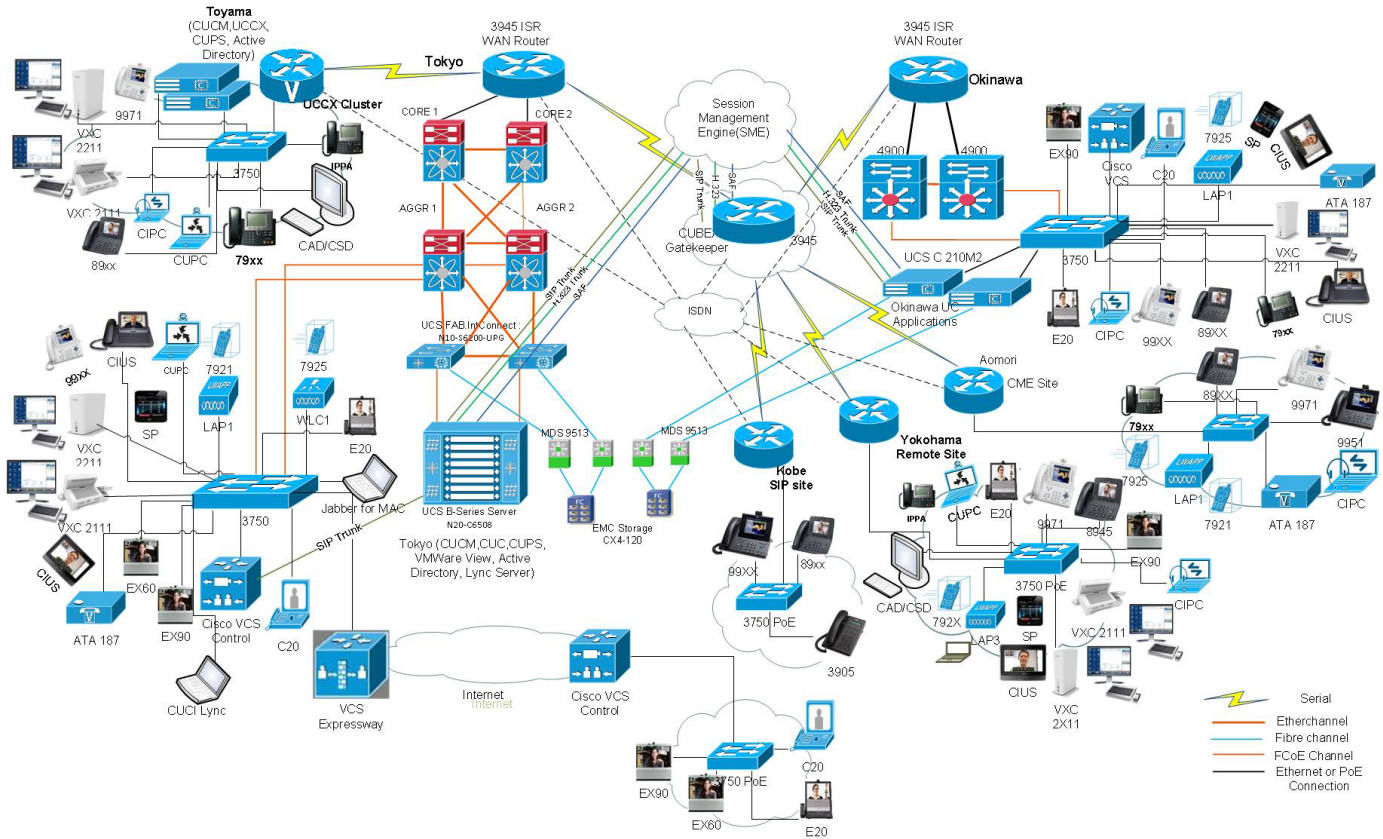
Test Topology and Environment

This chapter gives information on the following sections:

- [Test Topology](#)
- [Environment Matrix](#)
- [What's New?](#)
- [Open Caveats](#)

Test Topology

Figure 2-1 UC Solution Topology



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Environment Matrix

| Category | Component | Version | Remarks |
|--|---|--------------------------------|-----------|
| Call Control | Cisco Unified Communications Manager | 8.6.2.20000-2 | |
| | | 8.6.2.1000-1 | Locale |
| | | 3-1-9.JP | Dial Plan |
| | Cisco Unified Survivable Remote Site Telephony (SRST) | 15.2.2T | IOS |
| | Cisco Unified Communications Manager Express | 9 | |
| | | 15.2.2T | IOS |
| | | 8.8.2.4 | LI |
| | Cisco TelePresence Video Communication Server (VCS) | X7.0.3 | |
| Cisco TelePresence Video Communication Server Expressway | X6.1 | | |
| Applications | Cisco Unified Presence | 8.6.3.10000-20 | |
| | Cisco Unified Contact Center Express | 8.5.1 SU2 | |
| | Cisco Unity Connection | 8.6.2.21900-5 | |
| | | uc-locale-ja_JP-8.6.2.1-25.cop | Locale |

| Category | Component | Version | Remarks |
|----------------------------------|--|-----------------------------------|---------|
| Endpoints and Clients | E20 | TE4.1.1 | |
| | EX60 | TC5.1 | |
| | EX90 | TC5.1 | |
| | Cisco TelePresence Quick Set C20 | TC 5.1 | |
| | Cius | 9.2(2) | |
| | 6911 | SCCP69xx.9-2-2-5/SIP69xx.9-2-2-2 | |
| | 6921 | SCCP69xx.9-2-2-6/SIP69xx.9-2-2-7 | |
| | 6941 | SCCP69xx.9-2-2-6/SIP69xx.9-2-2-7 | |
| | 6945 | SCCP69xx.9-2-2-4/SIP69xx.9-2-2-6 | |
| | 6961 | SCCP69xx.9-2-2-6/SIP69xx.9-2-2-7 | |
| | 7961G | SCCP79xx.9-2-3/SIP79xx.9-2-3 | |
| | 7962 | SCCP79xx.9-2-3/SIP79xx.9-2-3 | |
| | 7965 | SCCP79xx.9-2-3/SIP79xx.9-2-3 | |
| | 7975 | SCCP79xx.9-2-3/SIP79xx.9-2-3 | |
| | 7985 | cmterm_7985.4-1-7-0 | |
| | 3905 | 9.2.2 | |
| | 3911 | 8.1.4(a) | |
| | 8941 | SCCP.89xx.9-2-3-5/SIP89xx.9-2-3-5 | |
| | 8945 | SCCP.89xx.9-2-3-5/SIP89xx.9-2-3-5 | |
| | 8961 | sip8961.9-2-3-27 | |
| | 9951 | sip9951.9-2-3-27 | |
| | 9971 | sip9971.9-2-3-27 | |
| | 7925G | CP7925G-1.4.2 | |
| | 7921G | CP7921G-1.4.2 | |
| | 7937G | apps37sccp.1-4-4-0 | |
| | Cisco IP Communicator | cipc-Admin-ffr.8-6-1-0 | |
| | CAD, CSD, CAD-BE | 8.5.1 | |
| | ATA 187 | 9.2.3 | |
| | Cisco Unified Personal Communicator | 8.5(5) | |
| | Cisco Unified Personal Communicator on MAC | 7.1.2 | |
| | Cisco Jabber for Mac | 8.6.2.18515 | |
| | Cisco Jabber for iphone | 8.6.3 | |
| Cisco IOS Voice and Data Gateway | 15.2.2T | | |
| Server | Microsoft Exchange Server | Exchange Server 2010 (JP-32 bit) | |

| Category | Component | Version | Remarks |
|---------------------------------|---|--------------------|---------|
| Client | OS | Win-XP Japanese | |
| | Cisco Virtualization Experience Client 2211 | 3.4(1) | |
| | Cisco Virtualization Experience Client 2111 | 3.4(1) | |
| | OS | Win 7 Japanese | |
| | Browser | IE 7 | |
| Fabric Interconnect PRIMARY | Cisco UCS 6100 | UCS Manager2.0(1t) | |
| Fabric Interconnect SUBORDINATE | Cisco UCS 6100 | UCS Manager2.0(1t) | |
| Fabric Cluster | Cisco UCS 6100 | UCS Manager2.0(1t) | |
| VMware View | VMware-viewagent | 4.6.0 | |
| VMware View | VMware-viewclient | 4.6.0 | |
| VMware View | VMware-viewconnectionserver | 4.6.0 | |
| Blade Server | ESXi | 4.1 | |
| VCenter Server | ESX | 4.1 | |
| Nexus | Nexus 1kV | 4.2(1)sp1.4 | |
| MDS Switch | M9500 | 5.2(2a) | |
| Wireless | | | |
| Controller | Wireless LAN controller 4404, 5508 | 7.2.1.79 | |
| | Wireless Flex Controller | 7.2.1.79 | |
| Access Point | Cisco AP 1200,1142,1042 3500 | 12.4 | |
| Client | OS | XP, windows 7 | |
| Browsers | IE, Mozilla, Chrome | 8,6.0.2,16 | |

What's New?

New Components:

- Cisco TelePresence Video Communication server (Cisco VCS)
- Cisco TelePresence Video Communication Server Expressway
- Cisco IP Video Phone E20
- Cisco TelePresence System EX90
- Cisco TelePresence System EX60
- Cisco TelePresence System Quick Set C20
- Cisco Cius
- Cisco Jabber for MAC
- Cisco Jabber for iPhone
- Cisco Unified IP Phone 894X
- Cisco Virtualization Experience Client 2111
- Cisco Virtualization Experience Client 2211



Note

The latest firmware of Cisco Jabber for iPhone is available at the App Store

The following table describes the new features in Cisco Unified Communications System release 8.6(2a):

Table 2-1 **New Features**

| New Feature | Description |
|---|---|
| CUPC 8.5.5 Interoperability between CUPC and Tandberg endpoints | Cisco Unified Personal Communicator now supports full interoperability with HD video-capable Tandberg endpoints. |
| Global Call park | Call Park codes are managed across the cluster by a single entity. |
| New Functionality in Disaster Recovery System | DRS have two added functionalities : <ul style="list-style-type: none"> • Estimation of backup tar size • DRS became intelligent enough to detect SFTP failure or space Issue |

Open Caveats

Open caveats describe the possible unexpected behaviors that you may encounter in release 8.6(2a) of Cisco Unified Communications System

| Defect ID | Defect Title |
|------------|--|
| CSCtt47002 | External Phone number mask not displays on 7937 IPPhone in srst |
| CSCtt97885 | after enabling CfwdAll in 6941 IP Phone it is not displaying in Japanese |
| CSCtt97831 | forwarding call with CfwdAll feature through Trunk |
| CSCtw27706 | QSIG enabled trunk is not listing Properly in the Available Devices list |
| CSCtx31388 | Unable to set callforward manually in the E20 reg with CUCM |
| CSCtx26931 | TC not showing referred by on an incoming transferred call |
| CSCty16844 | Cius not send call to remote dest by Mobility softky in japane |
| CSCty05174 | Transfer option missing in EX90(registered as H.323)in VCS. |
| CSCty11085 | Blurred video image while Calling From EX60 to 7985 Video Phone |
| CSCtx57070 | Pattern Numbers are not Stripped in Cisco TE EX60. |
| CSCtx60877 | Conference is displaying in E20 after conference over |
| CSCtx85121 | Conference is not Displaying in Cisco Jabber on iPhone |
| CSCty18933 | Transfer Issue on EX60 |

Known Issues

| Defect ID | Defect Title |
|------------|---|
| CSCtq71383 | Call coming to 8945 and 8941 IP Phones through CFwdAll Feature |
| CSCtw83213 | name change for desk phone control in Unified Presence GUI |
| CSCty48851 | Transfer: H323 endpoint as transferee when transferrer is registered to SIP |



CHAPTER 3

Test Results Summary

This chapter contains the following sections:

- [Cisco Unified Communications Manager](#)
- [Cisco TelePresence Video Solutions](#)
- [Cisco Virtualization Experience Client](#)
- [Cisco Jabber for iPhone](#)
- [Cisco Jabber for Mac](#)
- [Cisco Cius](#)
- [Cisco Unified Border Element](#)
- [Cisco Unified Contact Center Express](#)
- [Cisco Unified Presence](#)
- [Cisco Unity Connection](#)
- [Cisco UC Integration™ for Microsoft LYNC](#)
- [Cisco Unified Personal Communicator](#)
- [Cisco Unified Personal Communicator on Mac](#)
- [Cisco Unified Communications Manager Express](#)
- [Regression](#)
- [Related Documentation](#)

Cisco Unified Communications Manager

| Logical ID | Title | Description | Call Component Flow | Status | Defects |
|-------------------------|---|---|---------------------|--------|---------|
| UCJ862S.CUC M.SR.115 | Ringling sound for 6941 IP Phone (SIP) when selected default ringtone (chirp1, chirp2) | Verify that user is able to listen ringing sound on 6941 IP Phone when selected default ringtones (chirp1, chirp2) | | Passed | |
| UCJ862S.CUC M.SR.119 | Ringling sound for 6945 IP Phone (SIP) when selected default ringtone (chirp1, chirp2) | Verify that user is able to listen ringing sound on 6945 IP Phone when selected default ringtones (chirp1, chirp2) | | Passed | |
| UCJ862S.CUC M.SR.120 | Ringling sound for 6945 IP Phone (SCCP) when selected default ringtone (chirp1, chirp2) | Verify that user is able to listen ringing sound on 6945 IP Phone when selected default ringtones (chirp1, chirp2) | | Passed | |
| UCJ862S.CUC M.SR.123 | Ringling sound for 8945 IP Phone (SIP) when selected default ringtone(chirp1, chirp2) | Verify that user is able to listen ringing sound on 8945 IP Phone when selected default ringtones(chirp1, chirp2) | | Passed | |
| UCJ862S.CUC M.SR.124 | Ringling sound for 8945 IP Phone (SCCP) when selected default ringtone(chirp1,chirp 2) | Verify that user is able to listen ringing sound on 8945 IP Phone when selected default ringtones(chirp1,chirp2) | | Passed | |
| UCJ862S.CUC M.SR.125 | Ringling sound for 9951/71 IP Phone when selected default ringtone (chirp1,chirp2) | Verify that user is able to listen ringing sound on 9951/71 IP Phone when selected default ringtones (chirp1, chirp2) | | Passed | |
| UCJ862S.CUC M.SR.126 | Hold button status in 6961 IP Phone (SIP) when pickup notification on screen | Verify that hold button works properly in 6961 IP Phone when pickup notification displays on screen | | Passed | |
| UCJ862S.CUC M.SR.127 | Hold button status in 6961 IP Phone (SCCP) when pickup notification on screen | Verify that hold button works properly in 6961 IP Phone when pickup notification displays on screen | | Passed | |
| UCJ862S.CUC M.SR.132 | Hold button status in 6945 IP Phone (SIP) when pickup notification on screen | Verify that hold button works properly in 6945 IP Phone when pickup notification displays on screen | | Passed | |

| Logical ID | Title | Description | Call Component Flow | Status | Defects |
|-------------------------|---|--|---|--------|---------|
| UCJ862S.CUC M.SR.133 | Hold button status in 6945 IP Phone (SCCP) when pickup notification on screen | Verify that hold button works properly in 6945 IP Phone when pickup notification displays on screen | | Passed | |
| UCJ862S.CUC M.SR.134 | Hold button status in 8945 IP Phone (SIP) when pickup notification on screen | Verify that hold button works properly in 8945 IP Phone when pickup notification displays on screen | | Passed | |
| UCJ862S.CUC M.SR.135 | Hold button status in 8945 IP Phone (SCCP) when pickup notification on screen | Verify that hold button works properly in 8945 IP Phone when pickup notification displays on screen | | Passed | |
| UCJ862S.CUC M.T.202 | Call to the user that is searched from corporate directory | Verify that call to the user searched from corporate directory is successful | IP Phone -> Corporate directory -> search user; call to the user | Passed | |
| UCJ862S.CUC M.T.203 | Login to Extension Mobility on 8941 Unified IP Phone | Verify that an Extension Mobility user with Japanese locale could login into a Unified IP Phone 8941 with English locale | | Passed | |
| UCJ862S.CUC M.T.207 | Park reversion after customized reversion timer expires | Verify park reversion is successful after customized reversion timer expires | | Passed | |
| UCJ862S.CUC M.T.208 | Unified IP Phone 6961 login to Extension mobility through service URL | Verify that unified IP Phone 6961 login to Extension mobility through service URL successfully | | Passed | |
| UCJ862S.CUC M.T.210 | Two way audio call between Cisco Jabber for Mac and CUPC through SIP Trunk | Verify that two way audio call is established successfully between Cisco Jabber on Mac and CUPC through SIP Trunk. | | Passed | |
| UCJ862S.CUC M.G.162 | Cisco unified IP phone 7975 barge into an active call made by EMCC user | Verify that the Cisco EMCC user logging into the visiting cluster, initiates the call to Cisco unified IP phone 6945 and Cisco unified IP phone 7975 barge the call successfully | 6921 -> site A unified CM -> Sip trunk -> site B unified CM -> 6945(sip); 7975 -> barge | Passed | |

| Logical ID | Title | Description | Call Component Flow | Status | Defects |
|------------------------|---|--|---|--------|---------|
| UCJ862S.CUC M.G.156 | EMCC user transfers the incoming call. | Verify that Extension Mobility Cross Cluster feature enabled user (Unified IP Phone 9951) login to Visiting cluster and transfers the incoming call. | 6941(sip) -> site B unified CM -> Sip trunk -> site A unified CM -> 9951 ; 9951 -> call transfer -> site A unified CM -> 6945(sip) | Passed | |
| UCJ862S.CUC M.G.167 | Call back feature works with Cisco Extension mobility cross cluster user | Verify that Call back feature works successfully on a Cisco Extension mobility cross cluster user (Cisco Unified IP Phone 9951) | 9951 -> site A unified CM -> sip trunk -> site B unified CM -> 9971 | Passed | |
| UCJ862S.CUC M.G.160 | Cisco Unified IP Phone 6941 park the call from EMCC user (Cisco Unified IP Phone 9951) | Verify that the Cisco unified IP phone 6941(sccp) parks the call from EMCC user (Cisco Unified IP Phone 9951) successfully | 9951-> site A unified CM ->Sip trunk -> site B unified CM ->6941 ; 6941 ->call park -> 6945(sccp) retrieves parked call | Passed | |
| UCJ862S.CUC M.G.144 | Cisco unified IP phone 6941(sip) hold the call from EMCC user (Cisco Unified IP Phone 9951) | Verify that Cisco unified IP phone 6941(sip) hold and resume the call from EMCC user (Cisco Unified IP Phone 9951) successfully | 9951 ->site A unified CM-> SIP trunk-> site B unified CM-> 6941(sip) holds and resume | Passed | |
| UCJ862S.CUC M.G.152 | Call forward all on EMCC user (Cisco Unified IP Phone 9951) | Enable Call forward all on EMCC user (Cisco Unified IP Phone 9951). Make a call to EMCC user. Verify that call gets forwarded successfully. | 9971 -> site B unified CM -> SIP trunk -> site A unified CM -> 9951 ; 9951 -> call forward all -> 8961 | Passed | |
| UCJ862S.CUC M.G.165 | Call forward busy on EMCC user (Cisco Unified IP Phone 9951) | Enable Call forward busy on EMCC user (Cisco Unified IP Phone 9951). Make a call to EMCC user which is busy on other call. Verify that call gets forwarded successfully. | 6941(sip) -> site B unified CM -> Sip trunk -> site A unified CM -> 9951 ; 6945(sip) -> site A unified CM -> 9951 ; 9951 -> forward -> 9971 | Passed | |

| Logical ID | Title | Description | Call Component Flow | Status | Defects |
|------------------------|--|--|---|--------|---------|
| UCJ862S.CUC M.G.171 | Call forward no answer on EMCC user (Cisco Unified IP Phone 9951) | Enable Call forward no answer and Do not Disturb on EMCC user (Cisco Unified IP Phone 9951). Make a call to EMCC user. Verify that call gets forwarded successfully. | 9971-> site A unified CM -> 9951 ; 9951 ->forwards ->site A unified CM ->sip trunk ->site B unified CM-> 6945 | Passed | |
| UCJ862S.CUC M.G.182 | Cisco Unified IP Phone 7965 join the call from EMCC user (Cisco Unified IP Phone 9951) and Cisco Unified IP Phone 7921 | Verify that Cisco Unified IP Phone 7965 joins the call from EMCC user (Cisco Unified IP Phone 9951) and Cisco Unified IP Phone 7921 successfully | 9951 -> 7965 ; 7921(wireless) -> 7965 ; 7965->joins | Passed | |
| UCJ862S.CUC M.G.154 | Cisco Unified IP Phone 7975 join the call from two EMCC user (Cisco Unified IP Phone 9951 and Cisco Unified IP Phone 6921) | Verify that Cisco Unified IP Phone 7975 join the calls from two EMCC users (Cisco Unified IP Phone 9951 and Cisco Unified IP Phone 6921) successfully | 6921(EMCC user) -> site A unified CM -> 7975 ; 9951(EMCC user) -> 7975->attends and joins | Passed | |
| UCJ862S.CUC M.G.168 | Consult transfer made by Cisco Extension mobility cross cluster user | Verify that EMCC user (Cisco Unified IP Phone 9951) Consult transfer the call successfully | 9971 -> site A unified CM -> 9951 ; 9951 ->consult transfer -> 8961 | Passed | |
| UCJ862S.CUC M.G.155 | Cisco Unified IP Phone 7975 join the call from EMCC user (Cisco Unified IP Phone 6921) through SIP Trunk | Verify that Cisco Unified IP Phone 7975 join the call from EMCC user (Cisco Unified IP Phone 6921) through SIP Trunk successfully | 6945(sip) -> site B unified CM -> Sip trunk ->site A unified CM ->7975 ; 6921 -> 7975 ->attends and joins | Passed | |
| UCJ862S.CUC M.G.170 | EMCC user call to a Hunt pilot | EMCC user (Cisco Unified IP Phone 9951) calls to a Hunt pilot number. Verify that Line numbers associated with the hunt pilot rings successfully. | | Passed | |

| Logical ID | Title | Description | Call Component Flow | Status | Defects |
|------------------------|---|--|---|--------|---------|
| UCJ862S.CUC M.G.181 | Transfer call to a EMCC user | Verify that a call from a Cisco Unified Communications Manager Express gets transferred to a EMCC user successfully. | 6945 -> unified CME -> sip trunk -> site A unified CM-> 9971; 9971 ->transfer -> 6921 (EMCC user) | Passed | |
| UCJ862S.CUC M.G.176 | Japanese EMCC user login into Cisco unified IP phone 6921(English) | Verify that the Japanese EMCC user login to Cisco unified IP phone 6921 with english locale successfully | | Passed | |
| UCJ862S.CUC M.G.179 | Calls between Cisco unified IP phone 9951(EMCC user) and Cisco unified IP phone 7921 (wireless) | Verify that EMCC user (Cisco unified IP phone 9951) successfully makes call to Cisco unified IP phone 7921 | 9951 ->site A unified CM -> 7921 | Passed | |
| UCJ862S.CUC M.G.141 | EMCC user (Cisco Unified IP Phone 9951) conference | Verify that the Cisco EMCC user login to visiting cluster can make conference successfully | 9951 -> site A unified CM->SIP trunk->site B unified CM->6941 ; 9951 ->site A unified CM->SIP trunk ->site B unified CM -> 6945 (conference) | Passed | |
| UCJ862S.CUC M.G.184 | Cisco Unified IP Phone 7921 retrieves the parked call | Verify that the Cisco Unified IP Phone 7921 (wireless) successfully retrieves the parked call | 9951 -> site A unified CM -> 7965 ->parks; 7921 IP phone (wireless) retrieve parked call | Passed | |
| UCJ862S.CUC M.G.164 | Cbarg into a call on EMCC user (Cisco Unified IP Phone 9951) | Verify that the Cisco EMCC user login into the visiting cluster initiates the call and Cisco unified IP phone 7975 cbarg the call successfully | 9951 -> site A unified CM -> Sip trunk -> site B unified CM-> 6945; 7975-> Cbarg -> 9951,6945,7975 conference | Passed | |

| Logical ID | Title | Description | Call Component Flow | Status | Defects |
|-------------------------|--|--|---|--------|------------|
| UCJ862S.CUC M.G.186 | Cisco unified IP Phone 6921(EMCC user) Hold and Resume the call from Cisco Cius | Verify that the Cisco unified IP Phone 6921 (EMCC user) successfully hold and resumes the call from Cisco Cius | Cisco Cius -> site A unified CM -> 6921-> hold and resume | Passed | |
| UC862F.CUC M.M.029 | External Phone number mask displays on 7937 Unified IP Phone in Unified SRST | Verify that 7937 Unified IP Phone displays External phone number mask successfully in Unified SRST mode | | Failed | CSCtt47002 |
| UCJ862F_SR.C UCM.057 | Japanese display status for CfwAll feature in 6941 IP Phone | Verify that user enables CfwAll feature in 6941 IP Phone and status on display shows correctly in Japanese | | Failed | CSCtt97885 |
| UCJ862F_SR.C UCM.058 | Forward call in 69XX IP Phone using CfwAll feature through trunk | Verify that call is forwarding successfully in 69XX IP Phone using CfwAll feature from one Call Manager to another call Manager through Trunk and display status on destination IP Phone shows correctly | | Failed | CSCtt97831 |
| UCJ862F_SR.C UCM.055 | QSIG enabled trunk is listing Properly in the Available Devices list when we add and remove the Non QSIG Trunk | Verify that the QSIG enabled trunk is listing Properly in the Available Devices list when we add and remove the Non QSIG Trunk | | Failed | CSCtw27706 |

Cisco Mobile Connect

| Logical ID | Title | Description | Call Component Flow | Status | Defects |
|------------------------|---|--|---|--------|---------|
| UCJ862S.CUC M.G.201 | Mobile Connect: Hold incoming call on the desktop phone, resume the call in shared line and send the call to remote device | Verify the following: Answer the incoming call on the desktop phone (Unified IP Phone) with shared line. Hold the call. Resume the call in the shared line. Send the call to the remote device using Mobility soft key on Unified IP phone 6900 series (SIP). | Unified IP Phone A -> Cisco Unified Communications Manager -> Unified IP Phone 6921, 6945, Remote device rings; Unified IP Phone 6921 -> answer -> Hold; Unified IP phone 6945 -> resume -> mobility -> remote device | Passed | |
| UCJ862S.CUC M.G.203 | Mobile Connect: Hold incoming call on the desktop phone (Unified IP Phone 9951), resume the call in shared line (Unified IP Phone 9971) and send the call to remote device | Verify the following: Answer the incoming call on the desktop phone in shared line. Hold the call. Resume the call in the shared line. Send the call to the remote device using Mobility soft key on Unified IP phone 9900 series. | Unified IP Phone A -> Cisco Unified Communications Manager -> Unified IP Phone 9951, 9971, Remote device rings; Unified IP Phone 9951 -> answer -> Hold; Unified IP Phone 9971 -> resume -> mobility -> remote device | Passed | |

| Logical ID | Title | Description | Call Component Flow | Status | Defects |
|------------------------|--|--|--|--------|---------|
| UCJ862S.CUC M.G.204 | Mobile Connect: Answer the incoming transferred call on desktop phone (Unified IP Phone 9900 series) and send the call to remote device | Verify the following: Desktop phone receives a transferred call. Answer the incoming call on the desktop phone. Send the call to the remote device using Mobility softkey on Unified IP phone 9900 series. | Unified IP Phone A -> Unified IP Phone B; Unified IP Phone B -> Transfer -> Unified IP Phone C; Unified IP Phone C (9951), Remote device rings; Unified IP Phone 9951 -> answer -> mobility -> remote device | Passed | |
| UCJ862S.CUC M.G.207 | Mobile Connect: Answer the incoming transferred call on desktop phone and send the call to remote device | Verify the following: Desktop phone receives a transferred call. Answer the incoming call on the desktop phone. Send the call to the remote device using Mobility soft key on Unified IP phone 6900 series (SIP). | Phone A -> Phone B; Phone B -> Transfer -> Phone C; Phone C (6921), Remote device rings; 6921 -> answer -> mobility -> remote device | Passed | |
| UCJ862S.CUC M.G.208 | Mobile Connect: Conference with remote device | Verify that remote device joins the conference successfully | Unified IP Phone A -> Unified IP Phone B; Unified IP Phone B -> conference -> Unified IP Phone C; Unified IP Phone C, Remote device rings; remote device -> answers; Unified IP Phone A, B, remote device -> conference. | Passed | |

| Logical ID | Title | Description | Call Component Flow | Status | Defects |
|------------------------|--|---|--|--------|---------|
| UCJ862S.CUC M.G.211 | Mobile Connect: Answer incoming call through SIP trunk on the remote device | Verify that Mobile connect enabled user can answer the incoming call through SIP trunk to a Cisco Unified IP Phone 9971 on a remote device | Unified IP Phone A -> Cisco Unified Communication s Manager A -> SIP Trunk -> Cisco Unified Communication s Manager B -> Unified IP Phone 9971, Remote device rings. | Passed | |
| UCJ862S.CUC M.G.215 | Mobile Connect: Answer incoming call through SIP trunk on the remote device | Verify that Mobile connect enabled user can answer the incoming call through SIP trunk to a Cisco Unified IP Phone 6900 series (SIP) on a remote device | Unified IP Phone A -> Cisco Unified Communication s Manager A -> SIP Trunk -> Cisco Unified Communication s Manager B -> Unified IP Phone 6921, Remote device rings. | Passed | |
| UCJ862S.CUC M.G.225 | Mobile Connect: Answer incoming call on the desktop phone (Unified IP Phone) and remote device stops ringing | Verify that using an user enabled with Mobile connect can answer the incoming call on Cisco Unified IP Phone 6900 series (SIP) and remote device stops ringing | Unified IP Phone A -> Cisco Unified Communication s Manager -> Unified IP Phone 6921, Remote device rings. | Passed | |
| UCJ862S.CUC M.G.226 | Mobile Connect: User add remote destination using Cisco Unified CM user options page | Verify that user could be able to add remote destination using Cisco Unified CM user options page successfully | | Passed | |
| UCJ862S.CUC M.G.227 | Mobile Connect: User add remote destination using Cisco Unified CM user options page and remote device rings | Verify that user could be able to add remote destination using Cisco Unified CM user options page successfully and remote device rings | | Passed | |

| Logical ID | Title | Description | Call Component Flow | Status | Defects |
|------------------------|---|---|---|--------|---------|
| UCJ862S.CUC M.G.228 | Mobile Connect: Remote destination with Do not Disturb enabled | Verify that remote device does not rings when Do not Disturb enabled on Remote destination | | Passed | |
| UCJ862S.CUC M.G.234 | Mobile Connect: Answer incoming call through ICT trunk on the remote device | Verify that Mobile connect enabled user can answer the incoming call through ICT trunk to a Cisco Unified IP Phone 9971 on a remote device | Unified IP Phone A -> Cisco Unified Communication s Manager A -> ICT Trunk -> Cisco Unified Communication s Manager B -> Unified IP Phone 9971, Remote device rings. | Passed | |
| UCJ862S.CUC M.G.239 | Mobile Connect: Park and retrieve the incoming call through SIP trunk. | Verify the following on Mobile connect enabled user: 1. Incoming call to desktop phone through SIP trunk, Park the call. 2. After reversion timer expires, desktop phone and remote device rings | Unified IP Phone A -> Cisco Unified Communication s Manager A -> SIP Trunk -> Cisco Unified Communication s Manager B -> desktop phone, Remote device rings; desktop phone -> park; park reversion -> desktop phone, Remote device rings. | Passed | |

| Logical ID | Title | Description | Call Component Flow | Status | Defects |
|------------------------|---|---|---|--------|---------|
| UCJ862S.CUC M.G.240 | Mobile Connect: Park and retrieve the incoming call through ICT trunk. | Verify the following on Mobile connect enabled user: 1. Incoming call to desktop phone through ICT trunk, Park the call. 2. After reversion timer expires, desktop phone and remote device rings | Unified IP Phone A -> Cisco Unified Communication s Manager A -> ICT Trunk -> Cisco Unified Communication s Manager B -> desktop phone, Remote device rings; desktop phone -> park; park reversion -> desktop phone, Remote device rings. | Passed | |
| UCJ862S.CUC M.G.241 | Mobile Connect: Conference with remote device and CUPC | Verify that CUPC, remote device joins the conference successfully | Phone A -> CUPC; CUPC -> conference -> Phone C; Phone C, Remote device rings; remote device -> answers; Phone A, CUPC, remote device -> conference. | Passed | |

Cisco TelePresence Video Solutions

Cisco TelePresence Video Communication Server

The Cisco TelePresence Video Communication Server (VCS) provides the most advanced telepresence and video conferencing call control in the industry. It enables any-to-any interoperability between all standards-compliant SIP and H.323 devices. VCS enables video communications devices to interoperate with unified communications and Voice over IP solutions, providing the end user the best experience possible no matter what device they are using to communicate.

Cisco TelePresence Video Communication Server Expressway

Cisco TelePresence Video Communication Server Expressway (Cisco VCS Expressway) opens the world outside the firewall to organizations using video communications. Cisco VCS Expressway enables business-to-business communications, empowers remote and home-based workers, and allows service providers to provide video communications to customers. Cisco VCS Expressway provides standards-based and secure firewall traversal for SIP and H.323 devices, and the Cisco VCS enables communication with the outside world regardless of SIP or H.323 protocol.

Cisco TelePresence System EX60 and EX90:

The Cisco TelePresence EX60 & EX90 transforms the workplace by combining work, communications, and collaboration - all on the desktop with just the touch of a finger. Colleagues can instantly work together whether they are separated by a hallway, a street, or several time zones. As part of the EX series, the Cisco TelePresence EX60 & EX90 supports vivid, lifelike 1080p30 video, natural collaboration, and the simplicity of a touch screen interface for a complete solution that helps ensure conversations are more productive.

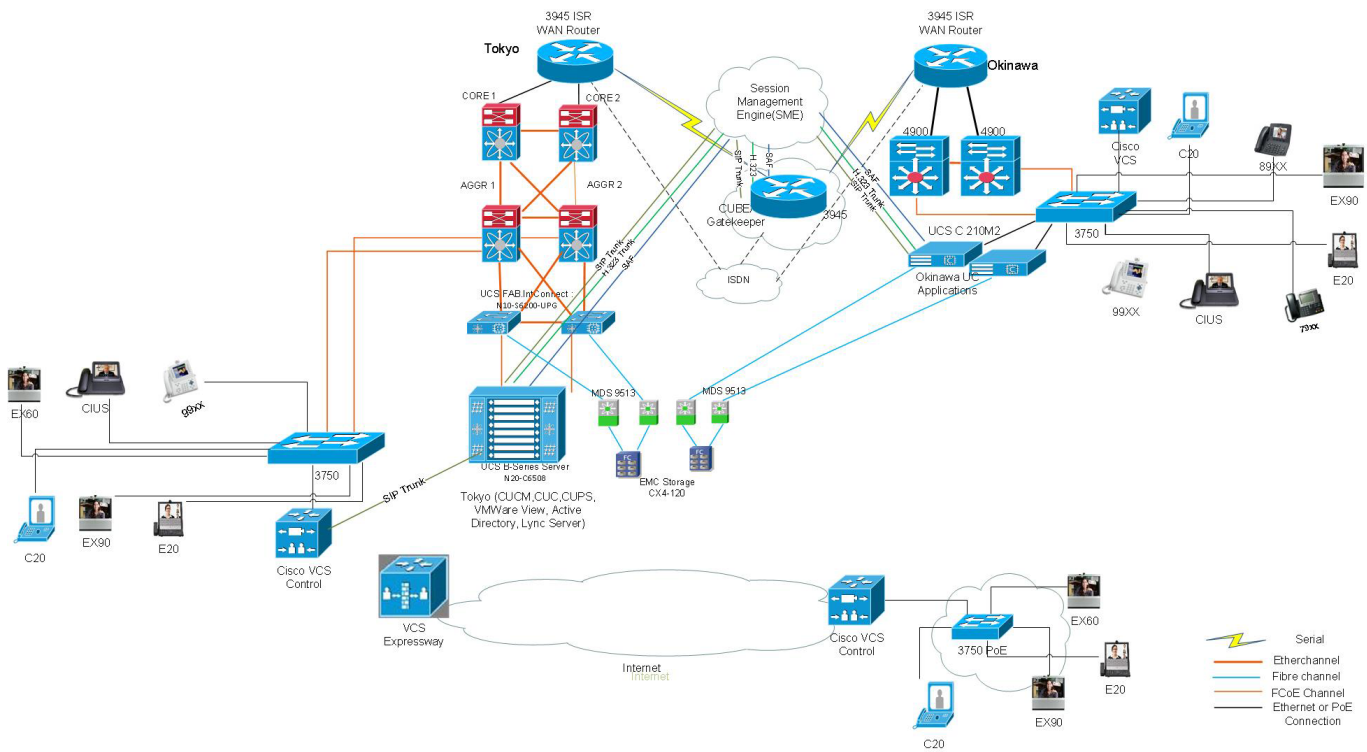
Cisco TelePresence System Quick Set C20:

It Transforms any meeting space into a video communications hub. The Quick Set C20 delivers the absolute quality of 1080p video in a simple-to-deploy, easy-to-manage and easy-to-use system. Whether you're just getting started with video communications or conducting a large-scale deployment, the Quick Set C20 delivers the performance you would expect from a larger system - in a compact, feature rich package. It is used for Video Conferencing and Presentation.

Cisco IP Video Phone E20

The Cisco E20 IP Video Phone Cisco is reinventing the desk phone by merging voice, video and collaboration into one device. A highly scalable solution for enterprise mass deployment, everyone in the organization will immediately see the benefits of increased productivity and daily collaboration.

Figure 3-1 Topology in Use



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| Logical ID | Title | Description | Call Component Flow | Status | Defect |
|--------------------|---|--|--|--------|--------|
| UCJ862S.Video .003 | Hold and Resume Video call Between Cisco TelePresence System EX90 and Cisco IP Video Phone E20 | Verify that hold and resume an active video call between EX90 and E20 works successfully | E20 -> Cisco VCS -> SIP Trunk ->Unified CM -> EX90 -> Hold and Resumes | Passed | |
| UCJ862S.Video .004 | Hold and Resume Video call Between Cisco TelePresence System EX60 and Cisco IP Video Phone E20 | Verify that hold and resume an active video call between EX60 and E20 works successfully | E20 -> Cisco VCS -> SIP Trunk ->Unified CM -> EX60 -> Hold and Resumes | Passed | |
| UCJ862S.Video .007 | Call Transfer Funtionality in Cisco IP Video Phone E20 | Verify when a call is transferred to E20, the caller ID should be correct in other E20. | E20->CISCO VCS->E20-> Call Transfer->EX60 | Passed | |
| UCJ862S.Video .009 | Call Transfer Funtionality in Cisco TelePresence System EX90 | Verify when a call is transferred to EX90 Phone , the caller ID should be correct in other EX90 Phone | EX90->CISCO VCS->EX90-> Call Transfer->E20 | Passed | |
| UCJ862S.Video .008 | Call Transfer Funtionality in Cisco TelePresence System EX60 | Verify when a call is transferred to EX60 Phone , the caller ID should be correct in other EX60 Phone | E20-> CISCO VCS -> E20-> Call Transfer->EX60 | Passed | |
| UCJ862S.Video .010 | Video call between Unified IP Phone 9971 and Cisco IP Video Phone E20 | Verify video call works successfully between 9971 (registered To Unified CM) and E20 (registered to CISCO VCS) | 9971->Unified CM->SIP Trunk-> Cisco VCS->E20 | Passed | |
| UCJ862S.Video .011 | Video call between Unified IP Phone 9951 (Unified CM) and CISCO VCS (Cisco IP Video Phone E20) | Verify video call works successfully between 9951 (registered To Unified CM) and E20 (registered to CISCO VCS) | 9971->Unified CM->SIP Trunk->Cisco VCS->E20 | Passed | |
| UCJ862S.Video .012 | Video call between Cisco IP Video Phone E20 (Unified CM) and CISCO VCS (Cisco IP Video Phone E20) | Verify video call works succesfully between E20 (registered to Unified CM) and E20 (registered to CISCO VCS) | E20->Unified CM->SIP Trunk->Cisco VCS->E20 | Passed | |

| Logical ID | Title | Description | Call Component Flow | Status | Defect |
|--------------------|--|---|--|--------|--------|
| UCJ862S.Video .013 | Video call between Unified CM (Cisco IP Video Phone E20) and CISCO VCS (Cisco TelePresence System EX60) | Verify video call works successfully E20 (registered To Unified CM) and EX60 (registered to CISCO VCS) | E20->Unified CM->SIP Trunk->Cisco VCS->EX60 | Passed | |
| UCJ862S.Video .014 | Video call between Unified CM (Cisco IP Video Phone E20) and CISCO VCS (Cisco TelePresence System EX90) | Verify video call can be established between E20 (registered To Unified CM) and EX90 (registered to CISCO VCS) | E20->Unified CM->SIP Trunk->Cisco VCS->EX90 | Passed | |
| UCJ862S.Video .015 | Make video call between Unified CM (Cisco TelePresence System EX60) and CISCO VCS (Cisco TelePresence System EX90) | Verify video call can be established between EX60 (registered To Unified CM) and EX90 (registered to CISCO VCS) | EX60->Unified CM->SIP Trunk->Cisco VCS->EX90 | Passed | |
| UCJ862S.Video .017 | Blind Transfer in Cisco TelePresence System EX60 | Verify whether able to attempt Blind Transfer in EX60 and transfer is successful. | EX60->CiscoVCS->EX60->Blind Transfer | Passed | |
| UCJ862S.Video .018 | Blind Transfer in Cisco IP Video Phone E20 | Verify whether able to attempt Blind Transfer in E20 and transfer is successful. | E20 -> CiscoVCS -> E20 -> Blind Transfer | Passed | |
| UCJ862S.Video .016 | Blind Transfer in Cisco TelePresence System EX90 | Verify whether Blind Transfer works Successfully in EX90 | EX90->CiscoVCS->EX90->Blind Transfer | Passed | |
| UCJ862S.Video .019 | Videocall between Sip Endpoint (Cisco IP Video Phone E20) and H323 Endpoint (Cisco TelePresence System EX60) | Verify whether video call works between Sip Endpoint(E20) and H323 Endpoint (EX60) in CISCO VCS. | E20(Sip) -> CiscoVCS -> EX60 (H.323) | Passed | |
| UCJ862S.Video .022 | Check Transfer Attended in Cisco TelePresence System EX90 | Verify whether the feature Transfer Attended works successfully in EX90 | EX90->CiscoVCS->EX90->Transfer Attended | Passed | |
| UCJ862S.Video .023 | Check Transfer Attended in Cisco TelePresence System EX60 | Verify whether the feature Transfer Attended works successfully in Cisco Video Phone EX60 | EX60->CiscoVCS->EX60->Transfer Attended | Passed | |

| Logical ID | Title | Description | Call Component Flow | Status | Defect |
|--------------------|---|--|---|--------|--------|
| UCJ862S.Video .024 | Check Transfer Attended in Cisco IP Video Phone E20 | Verify whether Transfer Attended works successfully in Cisco IP Video Phone E20 | E20->CiscoVCS->E20->Transfer Attended | Passed | |
| UCJ862S.Video .025 | Call Forward all in Cisco IP Video Phone E20 | Verify whether Call Forward all works successfully in E20 | E20 -> Unified CM -> E20 -> Call forward all | Passed | |
| UCJ862S.Video .026 | Call Forward all in Cisco TelePresence System EX60 | Verify whether the feature Call Forward all works successfully in EX60 | EX60->Unified CM->EX60->Call forward all | Passed | |
| UCJ862S.Video .031 | Video Call Between QUICK SET C20 Phone and Cisco TelePresence System EX60 | Verify video call works successfully between C20 and EX60 phones registered on CISCO VCS | QUICK SET C20 ->CISCO VCS->EX60 IP Phone | Passed | |
| UCJ862S.Video .032 | Video Call Between QUICK SET C20 Phones and Cisco TelePresence System EX90 registered to CISCO VCS | Verify video call works successfully between C20 and EX90 phones are registered on CISCO VCS | QUICK SET C20->CISCO VCS->EX90 | Passed | |
| UCJ862S.Video .033 | Video Call Between QUICK SET C20 Phones and Cisco IP Video Phone E20 Phones registered to CISCO VCS | Verify video call works successfully between C20 and E20 Phones registered on CISCO VCS | Quick Set C20-> Cisco VCS-> E20 | Passed | |
| UCJ862S.Video .035 | Video Call Between QUICK SET C20 Phones and Cisco TelePresence System EX90 | Verify video call can be established between QUICK SET C20 and EX90 series IP Phones. | QUICK SET C20 video Phone->Unified CM ->SIP Trunk->CISCO VCS->EX90 IP Phone | Passed | |
| UCJ862S.Video .039 | Presentation share between Cisco TelePresence System EX90 and QUICK SET C20 | Verify whether EX90 and QUICK SET C20 user registered to Unified CM can share presentation. | EX90->Unified CM ->C20->initiate presentation share | Passed | |

| Logical ID | Title | Description | Call Component Flow | Status | Defect |
|--------------------|--|---|---|--------|--------|
| UCJ862S.Video .037 | Hold and Resume the Inter cluster Video call on QUICK SET C20 IP Phone. | Verify that hold and resume an active video call between QUICK SET C20 and Unified IP Phone 9971 works successfully | C20-> Unified CM1 -> SIP Trunk ->Unified CM2 ->9971 IP Phone; QUICK SET C20 -> Hold and Resumes | Passed | |
| UCJ862S.Video .040 | Presentation share between Cisco TelePresence System EX60 and Cisco TelePresence System Quick Set C20 | Verify Presentation share works successfully between EX60 and Quick Set C20 | C20->Unified CM ->EX60->initiate presentation share | Passed | |
| UCJ862S.Video .041 | Presentation share between Cisco IP Video Phone E20 and QUICK SET C20 | Verify Presentation share works successfully between E20and Quick Set C20. | C20->Unified CM ->E20->initiate presentation share | Passed | |
| UCJ862S.Video .048 | Intercluster video call between QUICK SET C20 and Cisco TelePresence System EX90 | Verify Intercluster video call works between Quick Set C20 and EX90 Phones | C20 -> UnifiedCM1 -> SIPTrunk -> UnifiedCM2 -> EX90 | Passed | |
| UCJ862S.Video .049 | Intercluster video call between QUICK SET C20 and Cisco TelePresence System EX60 | Verify Intercluster video call works between C20 and EX60 IP Phones | C20->Unified CM1->SIP Trunk -> Unified CM2 ->EX60 | Passed | |
| UCJ862S.Video .050 | Intercluster video call between QUICK SET C20 and Cisco IP Video Phone E20 | Verify intercluster video call works between C20 and E20 | C20->Unified CM1->SIP Trunk -> Unified CM2 ->E20 | Passed | |
| UCJ862S.Video .055 | Video Conference between Cisco IP Video Phone E20, Cisco TelePresence System EX60, Cisco TelePresence System EX90 in CISCO VCS | Verify video conference works between E20, EX60 and EX90 registered to CISCO VCS | EX90->Cisco VCS->E20and EX60->Video conference | Passed | |

| Logical ID | Title | Description | Call Component Flow | Status | Defect |
|--------------------|--|---|---|--------|--------|
| UCJ862S.Video .058 | Video Conference between QUICK SET C20, Cisco TelePresence System EX60 , Cisco TelePresence System EX90 in CISCO VCS | Verify video conference works successfully between EX90, EX60 and C20 registered to CISCO VCS | EX60->Cisco VCS->C20 and EX90->Video conference | Passed | |
| UCJ862S.Video .059 | Video Conference between Cisco IP Video Phone E20, Cisco TelePresence System EX60 , Cisco TelePresence System EX90 in Unified CM | Verify video conference works successfully between EX90, EX60 and E20 registered to Unified CM | EX90->Unified CM->EX60and E20->Video conference | Passed | |
| UCJ862S.Video .061 | Intercluster Video Conference between Cisco IP Video Phone E20, Cisco TelePresence System EX60, Cisco TelePresence System EX90 in Unified CM | Verify video conference works successfully between EX90, E20 and EX60 registered to Unified CM | EX90 and EX60 ->Unified CM1->SIP Trunk->Unified CM2-> E20->Video conference | Passed | |
| UCJ862S.Video .062 | Intercluster Video Conference between QUICK SET C20, Cisco TelePresence System EX60 and Cisco TelePresence System EX90 in Unified CM | Check whether video conference between EX60, C20, EX90 in Unified CM is successful. | C20 and EX60->Unified CM1->SIP Trunk->Unified CM2->EX90-> Video conference | Passed | |
| UCJ862S.Video .066 | Video call between H.323 Endpoint (Cisco TelePresence System EX90)and SIP endpoint (Quick set C20) | Check whether video call works between H323 endpoint (EX90) and SIP Endpoint (C20) in CISCO VCS | EX90 (H.323 Endpoint)->Cisco VCS->C20 (SIP Endpoint) | Passed | |
| UCJ862S.Video .069 | Intercluster Video call between H.323 Endpoint (Cisco TelePresence System EX60) and SIP Endpoint (Quick set C20) | Check whether intercluster video call works between H323 (EX60) endpoint and SIP (Quick set C20) in CISCO VCS | EX60(H.323 Endpoint)-> Cisco VCS1->Cisco VCS2-> C20(SIP Endpoint) | Passed | |

| Logical ID | Title | Description | Call Component Flow | Status | Defect |
|--------------------|--|---|---|--------|------------|
| UCJ862S.Video .070 | Intercluster Video call between H.323 (Cisco IP Video Phone E20) and SIP Endpoint (Quick set C20) | Check whether video call works between H323 endpoint and SIP In CISCO VCS | E20(H.323 Endpoint)->Cisco VCS1->Cisco VCS2->C20(SIP Endpoint) | Passed | |
| UCJ862S.Video .071 | Video Conference between CiscoIP Video PhoneE20, Cisco TelePresence System EX60 and Cisco TelePresence System EX90 | Check whether able to perform video conference between Tandberg video phones in CISCO VCS | E20 and EX60->Unified CM->SIP Trunk->Cisco VCS->EX90->Video Conference | Passed | |
| UCJ862S.Video .072 | Video conferencing with different sites between between Cisco IP Video Phone E20 , Cisco TelePresence System EX60 and Cisco TelePresence System EX90 | Verify the Video Conferencing works successfully between EX90, EX60 and E20 | EX90->Unified CM1->SIP Trunk ->CiscoVCS->EX60 and E20->UnifiedCM2->Video Conference | Passed | |
| UCJ862S.Video .083 | Video Conference and Presentation share between Cisco IP Video Phone E20, Cisco TelePresence System EX60 and Cisco TelePresence System EX90 | Verify Video Conferencing works successfully between EX90, EX60 and E20 | EX60->Cisco VCS->E20and EX90->Video conference | Passed | |
| UCJ862S.Video .076 | Transfer call in Cisco TelePresence System EX60 when registered as H323 Endpoint | Check Call transfer is successful when EX60 registered as H323 Endpoint to Cisco VCS | EX60->Cisco VCS->Call Transfer | Failed | CSCty05142 |
| UCJ862S.Video .084 | Consultative transfer in Cisco IP Video Phone E20(H323 Endpoint) | Verify whether consultative Transfer is working in E20 (H323 Endpoint) | E20->Cisco VCS->E20->Consultative transfer | Failed | CSCty05243 |
| UCJ862S.Video .081 | Check Callforward manually in Cisco IP Video Phone E20 | Verify whether CallForward is able to set manually in E20 Registered with Unified CM | E20->CUCM->E20->CallFwdall | Failed | CSCtx31388 |
| UCJ862S.Video .086 | Check Callforward manually in Cisco IP Video Phone E20 | Verify whether CallForward is able to set manually in E20 registered with Cisco VCS | E20->Cisco VCS->E20->CallFwdall | Failed | CSCtx31608 |

| Logical ID | Title | Description | Call Component Flow | Status | Defect |
|--------------------|---|--|---|--------|------------|
| UCJ862S.Video .087 | Check Referred number in Cisco TelePresence System EX60 and Cisco TelePresence System EX90 | Verify whether referred number is showing in EX60 and EX90 after Call Transfer | EX60->CUCM->EX90->Call Transfer | Failed | CSCtx26931 |
| UCJ862S.Video .088 | Check Image Quality in Cisco TelePresence System EX60 | Verify the video image is clear while calling from EX60 to 7985 Video Phone | EX60->CUCM->7985 | Failed | CSCty11085 |
| UCJ862S.Video .082 | Hold and Resume Video call Between Cisco TelePresence System EX90 and Cisco IP Video Phone E20 Cisco VCS Expressway | Verify that hold and resume an active video call between EX90 and E20 works successfully | EX90->Unified CM->Cisco VCS Expressway->E20 | Passed | |
| UCJ862S.Video .089 | Presentation Between Cisco TelePresence System EX90 IP Phone and Cisco IP Video Phone E20 Cisco VCS Expressway | Verify that Presentation EX90 between and E20 IP Phone works successfully | EX90->Unified CM->Cisco VCS Expressway->E20 | Passed | |
| UCJ862S.Video .090 | Call Transfer between Cisco TelePresence System EX60 IP Phone and Cisco TelePresence System EX90 Cisco VCS Expressway | Verify that Call Transfer between EX60 and EX90 works successfully | EX60->Unified CM->CiscoVCS Expressway->EX90 -> Call Transfer->E20 | Passed | |
| UCJ862S.Video .091 | Checking the Uncontrolled Warning message in Cisco VCS Expressway | Verify that Uncontrolled Shutdown Warning message Displays successfully | | Passed | |
| UCJ862S.Video .092 | Auto answer feature Cisco TelePresence System EX60 registered in SITE A through Cisco VCS Expressway | Verify that auto answer feature works successfully | EX90->SITE B-> VCS Expressway-> VCS Control->EX60 | Passed | |
| UCJ862S.Video .093 | Check Farend Control mode feature in Cisco TelePresence QUICKSETC20 registered in SITE A through Cisco VCS Expressway | Verify that Farend Control mode feature works Successfully | C20->SITE B-> VCS Expressway-> VCS Control->EX60 | Passed | |

| Logical ID | Title | Description | Call Component Flow | Status | Defect |
|--------------------|--|--|--|--------|------------|
| UCJ862S.Video .094 | Video Call Between Cisco IP Video Phone E20 Phones and Cisco TelePresence System EX90 Phones registered to Cisco VCS Control and Cisco VCS Expressway | Verify video call can be established between E20 and EX90 IP Phones registered to Cisco VCS Control and Cisco VCS Expressway | E20->Cisco VCS->->Cisco VCS Expressway->EX90 | Passed | |
| UCJ862S.Video .095 | Video Conferencing Between Cisco IP Video Phone E20, Cisco TelePresence System EX60 Phones and Cisco TelePresence System EX90 Phones registered to Cisco VCS Control and Cisco VCS Expressway | Verify video call can be established between E20, EX90 and EX60 IP Phones registered to Cisco VCS Control and Cisco VCS Expressway | E20->EX60->Cisco VCS->Cisco VCS Expressway->EX90 | Passed | |
| UCJ862S.Video .096 | Video Conferencing - Presentation Between Cisco IP Video Phone E20, Cisco TelePresence System EX60 Phones and Cisco TelePresence System EX90 Phones registered to Cisco VCS Control and Cisco VCS Expressway | Verify video call can be established between E20, EX90 and EX60 IP Phones registered to Cisco VCS Control and Cisco VCS Expressway | E20->EX60->Cisco VCS->Cisco VCS Expressway->EX90 | Passed | |
| UCJ862S.Video .097 | Call Transfer Feature working registered in VCS as H.323 | Verify Call Transfer Feature is working | EX90->Cisco VCS Expressway->EX60 | Failed | CSCty05174 |
| UCJ862S.Video .098 | Check Presence Status of Cisco TelePresence System EX60 registered in SITE A through VCS Expressway | Verify that Presence Status works Successfully | EX90->Cisco VCS Expressway->EX60 | Passed | |

| Logical ID | Title | Description | Call Component Flow | Status | Defect |
|--------------------|---|---|---|--------|--------|
| UCJ862S.Video .099 | Check Publishers of Cisco TelePresence System EX60 registered in SITE A through Cisco VCS Expressway | Verify that Publishers Status is displayed in Cisco VCS Expressway | EX90->SITE B-> VCS Expressway-> VCS Control->EX60 | Passed | |
| UCJ862S.Video .100 | Check Subscribers of Cisco TelePresence System EX60 registered in SITE A through Cisco VCS Expressway | Verify that Subscribers Status is displayed in Cisco VCS Expressway | EX90->SITE B-> Cisco VCS Expressway-> Cisco VCS Control->EX60 | Passed | |
| UCJ862S.Video .101 | Cisco VCS Expressway Data Backup | Verify that Cisco VCS Expressway is Backed up | | Passed | |
| UCJ862S.Video .102 | Cisco VCS Expressway data restore | Verify that Cisco VCS Expressway data is restored | | Passed | |
| UCJ862S.Video .103 | Video Conferencing Between Cisco IP Video Phone E20,Cisco TelePresence System EX60(H.323) Phones and Cisco TelePresence System EX90(SIP) Phones registered to Cisco VCS Control and Cisco VCS Expressway | Verify that video call can be established between E20, EX90 (SIP) and EX60 (H.323) IP Phones registered to Cisco VCS Control and Cisco VCS Expressway | E20->EX60->Cisco VCS->Cisco VCS Expressway->EX90 | Passed | |
| UCJ862S.Video .104 | Video Conferencing Between Cisco IP Video Phone E20, Cisco TelePresence System EX60(SIP) Phones and Cisco TelePresence System EX90(H.323) Phones registered to Cisco VCS Control and Cisco VCS Expressway | Verify video call can be established between E20,EX90 (H.323) and EX60 (SIP) IP Phones registered to Cisco VCS Control and Cisco VCS Expressway | E20->EX60->Cisco VCS->Cisco VCS Expressway->EX90 | Passed | |

| Logical ID | Title | Description | Call Component Flow | Status | Defect |
|-----------------------|--|--|---|--------|--------|
| UCJ862S.Video .105 | Video Conferencing Between Cisco IP Video Phone E20,Cisco TelePresence System EX60(H.323) Phones and Cisco TelePresence System EX90(H.323) Phones registered to Cisco VCS Control and Cisco VCS Expressway | Verify video call can be established between E20, EX90 (H.323) and EX60 (H.323) IP Phones registered to Cisco VCS Control and Cisco VCS Expressway | E20->EX60->C isco VCS->Cisco VCS Expressway->E X90 | Passed | |
| UCJ862S.Video .106 | Video Conferencing Between Cisco IP Video Phone E20,Cisco TelePresence System EX60(SIP) Phones and Cisco TelePresence System EX90(SIP) Phones registered to Cisco VCS Control and Cisco VCS Expressway | Verify video call can be established between E20, EX90 (SIP) and EX60 (SIP) IP Phones registered to Cisco VCS Control and Cisco VCS Expressway | E20->EX60->C isco VCS->Cisco VCS Expressway->E X90 | Passed | |
| UCJ862S.Video .107 | Video Conferencing Between Quickset C20(SIP),Cisco TelePresence System EX60(H.323) Phones and Cisco TelePresence System EX90(SIP) Phones registered to Cisco VCS Control and Cisco VCS Expressway | Verify video call can be established between Quickset C20, EX90(SIP) and EX60 (H.323) IP Phones registered to Cisco VCS Control and Cisco VCS Expressway | Quick SetC20->EX60- >Cisco VCS->Cisco VCS Expressway->E X90 | Passed | |

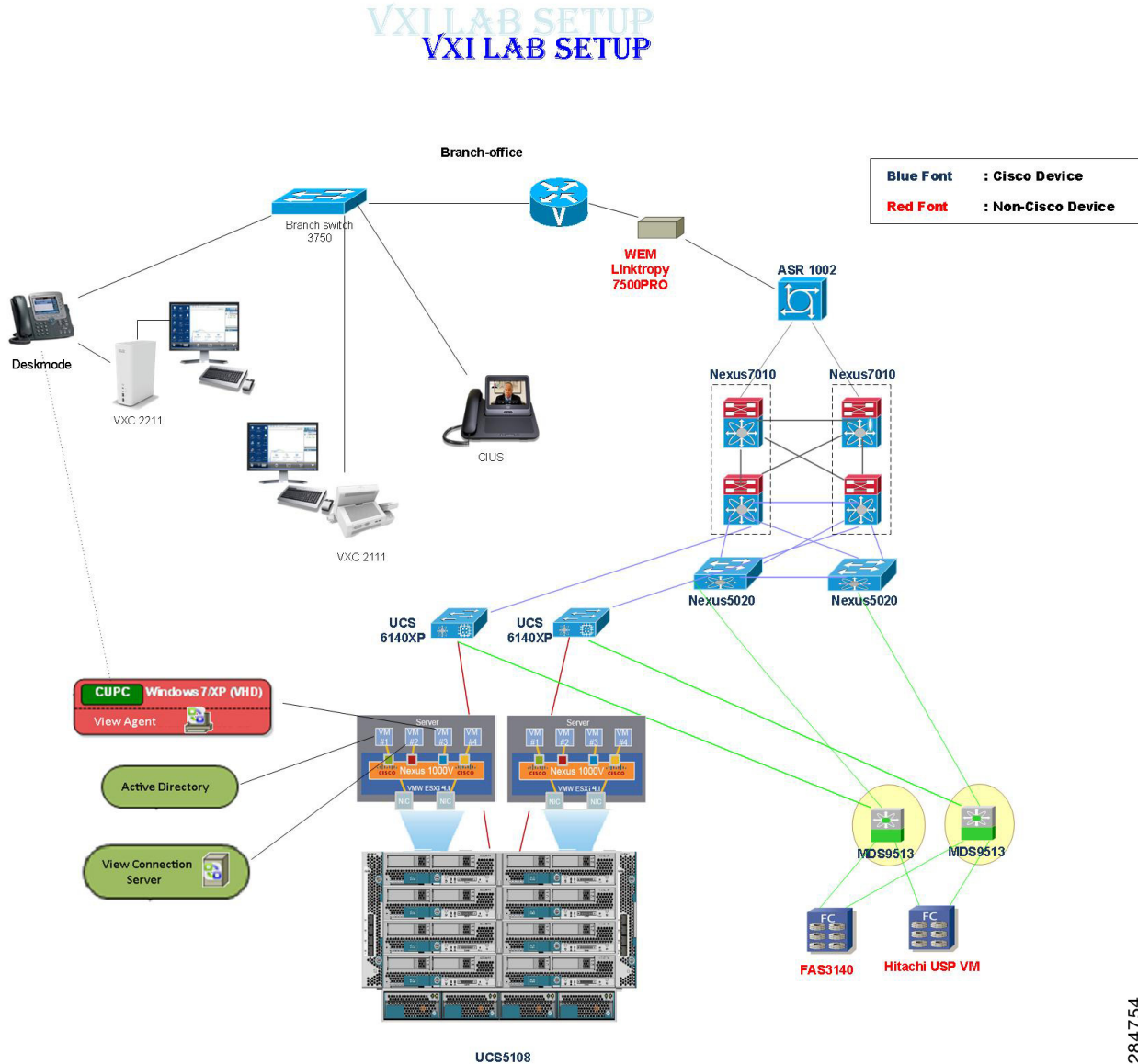
| Logical ID | Title | Description | Call Component Flow | Status | Defect |
|--------------------|--|---|---|--------|--------|
| UCJ862S.Video .108 | Video Conferencing Between Quickset C20(SIP),Cisco TelePresence System EX60(H.323) Phones and Cisco TelePresence System EX90(H.323) Phones registered to Cisco VCS Control and Cisco VCS Expressway | Verify video call can be established between Quickset C20, EX90 (H.323) and EX60 (H.323) IP Phones registered to Cisco VCS Control and Cisco VCS Expressway | Quick SetC20->EX60->Cisco VCS->Cisco VCS Expressway->EX90 | Passed | |
| UCJ862S.Video .109 | Video Conferencing Between Quickset C20(SIP),Cisco TelePresence System EX60(SIP) Phones and Cisco TelePresence System EX90(H.323) Phones registered to Cisco VCS Control and Cisco VCS Expressway | Verify video call can be established between Quickset C20, EX90 (H.323) and EX60 (SIP) IP Phones registered to Cisco VCS Control and Cisco VCS Expressway | Quick SetC20->EX60->Cisco VCS->Cisco VCS Expressway->EX90 | Passed | |
| UCJ862S.Video .110 | Video Conferencing - Presentation Between Cisco IP Video Phone E20, Cisco TelePresence System EX60(H.323) Phones and Cisco TelePresence System EX90(SIP) Phones registered to Cisco VCS Control and Cisco VCS Expressway | Verify video call can be established between E20, EX90 (SIP) and EX60 (H.323) IP Phones registered to Cisco VCS Control and Cisco VCS Expressway | E20->EX60->Cisco VCS->Cisco VCS Expressway->EX90 | Passed | |
| UCJ862S.Video .111 | Video Conferencing - Presentation Between Quick set C20, Cisco TelePresence System EX60 (H.323) Phones and Cisco TelePresence System EX90 (SIP) Phones registered to Cisco VCS Control and Cisco VCS Expressway | Verify video call can be established between Quick Set C20, EX90 (SIP) and EX60 (H.323) IP Phones registered to Cisco VCS Control and Cisco VCS Expressway | Quickset C20->EX60->Cisco VCS->Cisco VCS Expressway->EX90 | Passed | |

| Logical ID | Title | Description | Call Component Flow | Status | Defect |
|--------------------|--|---|---|--------|------------|
| UCJ862S.Video .112 | Video Conferencing - Presentation Between Quick set C20,Cisco TelePresence System EX60(H.323) Phones and Cisco TelePresence System EX90(H.323) Phones registered to Cisco VCS Control and Cisco VCS Expressway | Verify video call can be established between Quick Set C20,EX90(H.323) and EX60(H.323) IP Phones registered to Cisco VCS Control and Cisco VCS Expressway | Quickset C20->EX60->Cisco VCS->Cisco VCS Expressway->EX90 | Passed | |
| UC862S.CUC M.133 | Check Pattern Numbers are Stripped in Cisco TE EX60 Correctly. | Verify that the Cisco Unified Communications Manager handles the call and shared the information Correctly | EX60-->CUCM A-->Trunk -->SME -->Trunk-->CUCM B --> IP Phone B | Failed | CSCtx57070 |
| UC862S.CUC M.134 | Check the Status of the E20 after conference over | Verify that the Cisco Unified Communications Manager handles the call and shared the information Correctly | IP Phone A--->CUCM A--->E20-->Conference-->IP Phone C | Failed | CSCtx60877 |
| UC862S.CUC M.138 | Check for Transfer option on EX60 | To Verify that the Cisco Unified Communications Manager handles the call and shared the information Correctly | Cisco IP Phone A -> CUCM A -> SIP -> CUCM B -> EX60 | Failed | CSCty18933 |

Cisco Virtualization Experience Client

Cisco Virtualization Experience Infrastructure (VXI) is an end-to-end systems approach that delivers the next generation virtual workspace by unifying virtual desktops, voice, and video. The end-to-end systems approach offered by Cisco VXI delivers desktop virtualization and virtual workspaces that provide a superior desktop, voice and video user experience.

Figure 3-2 Topology in Use



284754

| Logical ID | Title | Description | Call Component Flow | Status | Defects |
|--------------------|--|--|---------------------|--------|---------|
| UCJ862S.Cius.G.107 | VXI: Cisco Cius access windows virtual desktop | Verify that Cisco Cius could access the Virtual Desktop (Japanese Windows 7) successfully | | Passed | |
| UCJ862S.Cius.G.108 | VXI: Cisco Cius access windows virtual desktop that is already login | Verify that Cisco Cius could access the Virtual Desktop (Japanese Windows 7) that is already login through any thick client. | | Passed | |
| UCJ862S.Cius.G.109 | VXI: Access windows virtual desktop from Thick client that is already login through Cisco Cius | Verify that Thick client could access the Virtual Desktop (Japanese Windows 7) that is already login through Cisco Cius. | | Passed | |
| UC862S.VXI.D.012 | Cisco Virtualization Experience Client 2111 native GUI sanity | Verify the basic GUI functionality of Cisco VXC 2111 | | Passed | |
| UC862S.VXI.D.013 | Cisco Virtualization Experience Client 2211 native GUI sanity | Verify the basic GUI functionality of Cisco VXC 2211 | | Passed | |
| UC862S.VXI.D.014 | Change of Presence status in Cisco Unified Personal Communicator | Verify Indication in message history window when remote user changes presence state to away. | | Passed | |
| UC862S.VXI.D.015 | Delete buddies from Contact list of Unified Personal Communicator | Verify that contact is deleted from the contact list successfully. | | Passed | |
| UC862S.VXI.D.016 | Add contact which is already added in the different group | Verify that contact gets added to a group which is already available in the different group successfully. | | Passed | |
| UC862S.VXI.D.018 | Cisco Unified Personal Communicator Music On Hold | Verify Unified Personal Communicator in deskphone mode hears Music On Hold, when the called party places the call on hold. | | Passed | |

| Logical ID | Title | Description | Call Component Flow | Status | Defects |
|--------------------|---|--|---|--------|---------|
| UC862S.VXI.D.019 | Blind Transfer the incoming to Cisco Unified Personal Communicator (deskphone mode) | Perform a blind call transfer using phone, then switch back to VM and verify Unified Personal Communicator GUI response | | Passed | |
| UC862S.VXI.D.020 | Group Chat | Perform a Group chat between the Unified Personal Communicator users. | | Passed | |
| UCJ862S.Cius.G.121 | VXI: Chat between two Unified Personal Communicators using Cius and Thick client | Verify that users are able to send/receive Instant messages between Unified Personal Communicators using Cius and Thick Client | | Passed | |
| UCJ862S.Cius.G.122 | VXI: Change the status for Unified Personal Communicator available in windows virtual desktop access through Cius | Verify that user is able to change the status for Unified Personal Communicator available in windows virtual desktop access through Cius | | Passed | |
| UCJ862S.Cius.G.123 | VXI: Edit the new status for Unified Personal Communicator available in windows virtual desktop access through Cius | Verify that user is able to edit the new status for Unified Personal Communicator available in windows virtual desktop access through Cius | | Passed | |
| UCCXJ851S.ACD.101 | Agent based routing | Verify the Agent based routing call | IP Phone ->GW->CUCM ->UCCX->CAD Agent | Passed | |
| UCCXJ851S.ACD.104 | CSQ based routing | Verify the CSQ based agent routing call | IP Phone ->GW->CUCM ->UCCX->CAD Agent | Passed | |
| UCCXJ851S.CAD.101 | Team message | Verify the team message | | Passed | |
| UCCXJ851S.CAD.102 | Call to another Agent from CAD | To verify make call in Cisco agent desktop | CAD agent A -> CAD agent B | Passed | |
| UCCXJ851S.CAD.103 | Agent status in supervisor desktop | Verify the agent status in supervisor desktop | | Passed | |
| UCCXJ851S.CAD.108 | Agent chat with Supervisor during active call | Verify Agent chat with Supervisor during an active call | | Passed | |

| Logical ID | Title | Description | Call Component Flow | Status | Defects |
|-----------------------|---|---|---|--------|---------|
| UCCXJ851S.C BE.102 | Verify the Conference Feature in CAD-BE | To verify the conference feature in CAD BE | PSTN Phone A -> GW -> CAD BE Agent -> Conference -> CSD | Passed | |
| UCCXJ851S.C BE.105 | Transfer the call to a supervisor from CAD-BE | To verify the transfer feature in CAD-BE | PSTN Phone A -> GW -> CAD BE Agent -> CSD | Passed | |
| UCCXJ851S.C BE.106 | CAD-BE Hold and Resume functionality | To verify the CAD-BE Hold and resume and functionality | | Passed | |
| UCCXJ851S.C BE.107 | CAD BE High priority chat to CSD | To verify the CAD BE High priority chat to CSD | | Passed | |
| UCCXJ851S.C SD.101 | Intercept feature in CSD | To verify the intercept feature in CSD | PSTN Phone A -> GW -> Agent -> CSD | Passed | |
| UCCXJ851S.C SD.102 | Monitoring Call from CSD | To verify that the CSD can monitor a call | | Passed | |
| UCCXJ851S.C SD.103 | Barge-in feature | To verify the barge-in feature | | Passed | |
| UCCXJ851S.C SD.104 | Forced log off from Cisco supervisor desktop | To verify the forced log off using Cisco supervisor desktop | | Passed | |
| UCCXJ851S.C SD.105 | Forced to ready state from CSD | To verify the corresponding state change from CSD | | Passed | |

Cisco Jabber for iPhone

| Logical ID | Title | Description | Call Component Flow | Status | Defects |
|-----------------------|--|--|---|--------|---------|
| UCJ862S.WIP. G.113 | Place a call from Cisco Jabber for iphone | Verify that the outgoing call from Cisco Jabber for iphone to corporate phone number is successful | Cisco Jabber for iphone ->Unified CM 1 ->SIP->Unified CM 2-> Unified IP phone A | Passed | |
| UCJ862S.WIP. G.114 | Receive a call to Cisco Jabber for iphone | Verify that the incoming call to Cisco Jabber for iphone from a corporate number is successful | Cius->Unified CM 1->SIP->Unifie d CM 2 ->Cisco Jabber for iphone | Passed | |
| UCJ862S.WIP. G.123 | Cisco Jabber for iphone- Deskphone mode | Verify that the deskphone mode in Cisco Jabber for iphone works successful | Unified IP Phone A->Unified CM 1->SIP->Unifie d CM 2->Unified IP Phone B and Cisco Jabber for iphone (Deskphone mode) both rings at the same time | Passed | |
| UCJ862S.WIP. G.126 | Directory Search and Caller Identification | Verify that the directory search in Japanese format on Cisco Jabber for iphone works successful | | Passed | |
| UCJ862S.WIP. G.127 | Integrating directory photo from LDAP server | Verify that directory photo is displayed on Cisco Jabber for iphone successfully | | Passed | |
| UCJ862S.WIP. G.179 | Cisco Jabber IM Single Resource Login | Verify that Single Resource login on Cisco Jabber IM for iphone works successfully | | Passed | |
| UCJ862S.WIP. G.181 | Cisco Jabber IM for iphone Click to IM | Verify that chat on Cisco Jabber IM for iphone works successfully | Cisco Jabber IM for iphone 1->Unified CM 1 ->Chat ->Cisco Jabber IM for iphone 2 | Passed | |

| Logical ID | Title | Description | Call Component Flow | Status | Defects |
|-----------------------|---|---|---|--------|---------|
| UCJ862S.WIP. G.185 | Cisco Jabber IM Add Contact | Verify that adding the contact on Cisco Jabber IM for iphone works successfully | | Passed | |
| UCJ862S.WIP. G.186 | Cisco Jabber IM Custom sizing the presence status | Verify that Presence status on Cisco Jabber IM for iphone works successfully | Cisco Jabber for iphone->Unified CM 1->Presence status->Unified Personal Communicator | Passed | |
| UCJ862S.WIP. G.187 | Cisco Jabber IM chat with Unified Personal Communicator | Verify the IM between the Unified personal Communicator and Cisco Jabber IM | Cisco Jabber IM for iphone ->Unified CM 1 ->Chat ->Unified Personal Communicator | Passed | |
| UCJ862S.WIP. G.188 | Cisco Jabber IM Chat with Cisco Jabber for MAC | Verify the IM between the Cisco Jabber for MAC and Cisco Jabber IM | Cisco Jabber IM for iphone->Unified CM 1 ->Chat ->Cisco Jabber for MAC | Passed | |
| UCJ862S.WIP. G.110 | Toggle between different clusters | Verify that the active scanning and calls does not fail while moving across clusters | Unified IP phone A ->Unified CM 1->Cisco Jabber for i phone | Passed | |
| UCJ862S.WIP. G.111 | SRST-Cisco Jabber for iphone | Verify that the Cisco Jabber for iPhone gets registered automatically to failover AP successfully | | Passed | |
| UCJ862S.WIP. G.112 | Call Via PSTN to Cisco Jabber for iphone | Verify that the call from PSTN to Cisco Jabber for iphone is successful | Unified IP phone A -> Unified CM 1 -> ISDN -> Remote Site ->Cisco Jabber for iphone | Passed | |

| Logical ID | Title | Description | Call Component Flow | Status | Defects |
|-----------------------|--|--|---|--------|------------|
| UCJ862S.WIP. G.120 | Swapping between two active calls | Verify that the Swap option in Cisco Jabber for iPhone works successfully | Unified IP phone A->Unified CM 1->Cisco Jabber for iPhone A Cisco Jabber for iPhone A(Connected)-> swap call ->Unified CM 1->SIP->Unified CM 2->Cius | Passed | |
| UCJ862S.WIP. G.128 | Enable SIP Digest Authentication with Automatic Credential entry | Verify that the security feature that authenticates device with automatic credential entry on Cisco Jabber for iPhone works successfully | | Passed | |
| UCJ862S.WIP. G.129 | Enable SIP Digest Authentication with Manual Credential entry | Verify that the security feature that authenticates device with manual credential entry on Cisco Jabber for iPhone works successfully | | Passed | |
| UC862S.CUC M.137 | Verify the Cisco Jabber on iPhone when it is in the Conference | To Verify that the Cisco Unified Communications Manager handles the call and shared the information correctly | Cisco-jabber on iPhone-->CUCM A-->H.323ICT -->CUCM B-->P phone A (conference)-->CUCM B -->IP phone B | Failed | CSCtx85121 |

Cisco Jabber for Mac

| Logical ID | Title | Description | Call Component Flow | Status | Defects |
|-------------------|---|--|--|--------|---------|
| UCJ862S.CJM.G.001 | Login to Cisco Jabber for Mac | Verify that Cisco Jabber for Mac user able to Login with their profile successfully | | Passed | |
| UCJ862S.CJM.G.002 | Status of Cisco jabber for Mac users | Verify the user can able to change the status Cisco Jabber for Mac user like Available, Busy, Custom, Away | | Passed | |
| UCJ862S.CJM.G.003 | Instant Messaging on Cisco Jabber for Mac | Verify that Instant Messaging between Cisco Jabber for Mac users (1:1 chat) is successful | | Passed | |
| UCJ862S.CJM.G.009 | Make an Audio call using Cisco Jabber for Mac | Verify that Cisco Jabber for Mac user login and made audio calls successfully | Soft phone A (Cisco Jabber for Mac) -> Unified CM -> Unified IP phone B | Passed | |
| UCJ862S.CJM.G.014 | Make an Audio call between two Cisco Jabber for Mac Users | Verify that Cisco Jabber for Mac calls to a Cisco Jabber for Mac successfully | Soft phone A (Cisco Jabber for Mac) -> Unified CM -> Soft phone B (Cisco Jabber for Mac) | Passed | |
| UCJ862S.CJM.G.021 | Make an Audio Call Conference using Cisco Jabber for Mac | Verify that Cisco Jabber for Mac user puts an audio call in Conference successfully | Unified IP phone A -> Unified CM -> Soft phone B (Cisco Jabber for Mac) -> Conference -> Unified CM -> Unified IP phone C | Passed | |
| UCJ862S.CJM.G.022 | Call Forward All using Cisco Jabber for Mac | Verify that Cisco Jabber for Mac user forwards the call using Call Forward All option successfully | Unified IP phone A -> Unified CM -> Soft phone B (Cisco Jabber for Mac) Call Forward All -> Unified CM -> Unified IP phone C | Passed | |

| Logical ID | Title | Description | Call Component Flow | Status | Defects |
|-----------------------|--|---|---|--------|---------|
| UCJ862S.CJM. G.023 | Call Park using Cisco Jabber for Mac | Verify that Cisco Jabber for Mac parks the call successfully | Unified IP phone A -> Unified CM -> Soft phone B (Cisco Jabber for Mac) -> Park; Unified IP phone C -> Retrieve | Passed | |
| UCJ862S.CJM. G.025 | Hold and Resume using Cisco Jabber for Mac | Verify that Cisco Jabber for Mac user Hold and resume the call successfully | Unified IP phone A -> Unified CM -> Soft phone B (Cisco Jabber for Mac) -> Hold and Resume | Passed | |
| UCJ862S.CJM. G.026 | Call Transfer using Cisco Jabber for Mac | Verify that Cisco Jabber for Mac user transfer the call successfully | Unified IP phone A -> Unified CM -> Cisco Jabber for Mac -> Transfer -> Unified IP phone C | Passed | |
| UCJ862S.CJM. G.031 | Sending/Receiving files using Cisco Jabber for Mac | Verifying Sending files between Cisco Jabber for Mac users is successful | | Passed | |

Cisco Cius

| Logical ID | Title | Description | Call Component Flow | Status | Defects |
|--------------------|---|---|--|--------|---------|
| UCJ862S.Cius.G.101 | Cisco Cius: Answer incoming call to Cisco Cius on the remote device | Verify that using Mobile connect feature an user can answer the incoming call to Cisco Cius on a remote device | Phone A -> Cisco Unified Communications Manager -> Cisco Cius, Remote device rings. | Passed | |
| UCJ862S.Cius.G.105 | Cisco Cius: Answer incoming call through ICT trunk on the remote device | Verify that Mobile connect enabled user can answer the incoming call through ICT trunk to Cisco Cius on a remote device | Phone A -> Cisco Unified Communications Manager A -> ICT Trunk -> Cisco Unified Communications Manager B -> Cisco Cius, Remote device rings. | Passed | |
| UCJ862S.Cius.G.106 | Cisco Cius: Answer incoming call through SIP trunk on the remote device | Verify that Mobile connect enabled user can answer the incoming call through SIP trunk to Cisco Cius on a remote device | Phone A -> Cisco Unified Communications Manager A -> SIP Trunk -> Cisco Unified Communications Manager B -> Cisco Cius, Remote device rings. | Passed | |
| UCJ862S.Cius.G.112 | CBarge the call between Cisco Cius and ATA 187 phone. | Cisco Cius makes call to ATA 187 through SIP trunk. Verify that 7965 phone charge into the call successfully | Cisco Cius -> Unified CM 1 -> SIP Trunk -> Unified CM 2 -> ATA 187; 7965 -> cbarge; Cisco Cius, ATA 187, 7965 -> conference | Passed | |

| Logical ID | Title | Description | Call Component Flow | Status | Defects |
|------------------------|---|---|---|--------|---------|
| UCJ862S.Cius. G.113 | Cisco Cius calls CUPC over SIP Trunk | Verify that Cisco Cius places call to CUPC over SIP trunk and CUPC transfers the call to unified IP Phone 9951. | Cisco Cius -> Cisco Unified Communications Manager A -> SIP Trunk -> Cisco Unified Communications Manager B -> CUPC -> Transfers -> Unified IP Phone 9951 | Passed | |
| UCJ862S.Cius. G.114 | Cisco Cius chat with Cisco Jabber on iPhone | Verify that Cisco Cius could chat with Cisco Jabber on iPhone successfully. | | Passed | |
| UCJ862S.Cius. G.116 | Multiple IM sessions on Cisco Cius with Cisco Jabber on iPhone and Cisco Jabber on Mac user | Begin an IM chat with a Cisco Jabber on iPhone user. While the chat is ongoing, initiate an IM chat with the Cisco Cius user from another Cisco Jabber on Mac client. Verify messages are properly exchanged between the two clients and that Cisco Cius is able to handle the multiple sessions. | | Passed | |
| UCJ862S.Cius. G.117 | Cisco Cius chat with Cisco Jabber on Mac user. | Begin an IM chat with a Cisco Jabber on Mac user. Verify messages are properly exchanged between the two clients. | | Passed | |
| UCJ862S.Cius. G.118 | Cisco Cius chat with Cisco Jabber on Mac user, multiple IM sessions | Begin an IM chat with a Cisco Jabber on Mac user. While the chat is ongoing, initiate an IM with the Cisco Cius user from another CUPC 8.5 client. Verify messages are properly exchanged between the two clients and that Cisco Cius is able to handle the multiple sessions. | | Passed | |

| Logical ID | Title | Description | Call Component Flow | Status | Defects |
|------------------------|---|---|---|--------|------------|
| UCJ862S.Cius. G.119 | Presence status update on Cisco Cius | Verify that change of presence status displays correctly on Cisco Cius and other presence contacts. | | Passed | |
| UCJ862S.Cius. G.120 | Escalate IM session to phone call | Begin an IM session with a user in another cluster. Escalate the IM session to a phone call. | | Passed | |
| UCJ862S.WIP. G.195 | Cisco Cius – Plus Dialing | Verify that + dialing works Successful | | Passed | |
| UCJ862S.WIP. G.196 | Cisco Cius - SRST | Verify that Cisco Cius registered with SRST on WAN failure | | | |
| UCJ862S.WIP. G.190 | Cisco Cius -Docking and Undocking a video call | Verify that device docking and undocking does not affects the Video call quality | Cisco Cius -> Unified IP Phone 8945; undock Cius | Passed | |
| UCJ862S.Cius. G.102 | Answer incoming call on Cisco Cius and send the call to remote device | Verify the following:Answer the incoming call on the Cisco Cius. Send the call to the remote device using Mobility softkey on Cisco Cius. | Phone A -> Cisco Unified Communications Manager -> Cisco Cius, Remote device rings; Cisco Cius -> answer -> mobility -> remote device | Failed | CSCty16844 |

Cisco Unified Border Element

| Logical ID | Title | Description | Call Component Flow | Status | Defects |
|------------------------|--|--|---|--------|---------|
| UCJ862S.CUB E.G.005 | Call Transfer by Cisco Unified IP phone 6921 to Cisco Cius in SIP to SIP interworking | Verify that the Call Transfer by Cisco Unified IP phone 6921 to Cisco Cius successfully in SIP to SIP interworking | Cisco Cius-> site A unified CM -> sip trunk ->cube ->sip trunk-> site B unified CM-> 6921 ; 6921-> transfer ->site B unified CM -> sip trunk ->cube ->sip trunk-> site A unified CM-> 9971 ; 6921-> transfer | passed | |
| UCJ862S.CUB E.G.011 | Call Transfer by Cisco Unified IP phone 9971 from Cisco Unified IP phone 7925(wireless) in SIP to SIP interworking | Verify that in SIP to SIP interworking the Call Transferred by Cisco Unified IP phone 9971 from Cisco Unified IP phone 7925(wireless) to Cisco Cius successfully | 7925(wireless) -> site A unified CM -> sip trunk ->cube ->sip trunk-> site B unified CM-> 9971 ; 9971-> transfer ->site B unified CM -> sip trunk ->cube ->sip trunk-> site A unified CM->Cisco Cius ; 9971-> transfer | passed | |
| UCJ862S.CUB E.G.006 | Conference by Cisco Unified IP phone 9951 in SIP to SIP interworking | Verify that the Conference made by Cisco Unified IP phone 9951 with Cisco Unified IP phone 7925(wireless) is successful in SIP to SIP interworking | 9951 -> site A unified CM -> sip trunk -> cube -> sip trunk -> site B unified CM -> 8945; Cisco Unified IP phone 9951 -> site A unified CM -> sip trunk -> cube -> sip trunk -> site B unified CM-> 7925(wireless); 9951-> conference | passed | |

| Logical ID | Title | Description | Call Component Flow | Status | Defects |
|------------------------|---|--|--|--------|---------|
| UCJ862S.CUB E.G.017 | Call join by Cisco Unified IP phone 7965 in SIP to SIP interworking | Verify that the Cisco Unified IP phone 7965 join the calls from Cisco Unified IP phone 8941 and 9971 successfully | 8941 -> site B unified CM -> sip trunk ->cube ->sip trunk-> site A unified CM-> 7965 ; 9971-> site B unified CM -> sip trunk ->cube ->sip trunk-> site A unified CM-> 7965 ; 7965-> join | passed | |
| UCJ862S.CUB E.G.049 | Cisco Unified IP phone 8945 hold and resume call from Cisco Cius | Verify that Cisco Unified IP phone 8945 hold and resume call from Cisco Cius successfully | Cisco Cius -> site A unified CM -> sip trunk ->cube ->sip trunk-> site B unified CM-> 8945 ; 8945 ->hold and resume | passed | |
| UCJ862S.CUB E.G.007 | Cisco Unified IP phone 6945 hold and resume call from Cisco Unified IP phone 7925(wireless) | Verify that Cisco Unified IP phone 6945 hold and resume call from Cisco Unified IP phone 7925(wireless) successfully | 7925(wireless)-> site A unified CM -> sip trunk ->cube ->sip trunk-> site B unified CM-> 6945 ; 6945 ->hold and resume | passed | |
| UCJ862S.CUB E.G.018 | Unicast MOH works with Cisco cube in SIP to SIP interworking | Verify that the Cisco unified IP phone 6941 receives the unicast music send by MOH server successfully | 6961 -> site B unified CM ->sip trunk ->cube ->sip trunk -> site A unified CM -> 6941 ; 6961 -> holds | passed | |

| Logical ID | Title | Description | Call Component Flow | Status | Defects |
|------------------------|--|---|---|--------|---------|
| UCJ862S.CUB E.G.022 | Conference made by Cisco unified IP phone 9971 in site A unified CM with Cisco unified IP phone 8941 in site B unified CM in SIP to SIP interworking | Verify that the conference call made by Cisco unified IP phone 9971 with Cisco unified IP phone 8941 in site B unified CM is successful | 9971 -> site A unified CM -> Cisco unified IP phone 9951 ; 9971 -> site A unified CM -> sip trunk -> cube -> sip trunk -> site B unified CM -> 8941; 9971 -> conference | passed | |
| UCJ862S.CUB E.G.014 | Call transfer made by Cisco unified IP phone 6945 to Cisco unified IP phone 7925 registered in unified CME | Verify that the transfer of call from Cisco unified IP phone 9951 in unified CM to Cisco unified IP phone 7925(wireless) in unified CME is successful | 9951 -> site A unified CM -> sip trunk -> cube -> sip trunk -> site B unified CM -> 6945 ; 6945-> transfer -> unified CME -> 7925(wireless) | passed | |
| UCJ862S.CUB E.G.001 | Call hold made by Cisco unified IP phone 8941 in SIP to SIP interworking | Verify that Cisco IP phone 8941 hold and resume the call successfully | 6941(sccp) ->site A unified CM ->sip trunk->cube ->sip trunk -> site B unified CM -> 8941 ; 8941 ->holds | passed | |
| UCJ862S.CUB E.G.052 | Call made by Cisco unified IP phone 89xx to Cisco unified IP phone 69xx(sip) in SIP to SIP interworking | Verify that the Call made by Cisco unified IP phone 89xx to Cisco unified IP phone 69xx(sip) in site B unified CM is successful | 8941 -> site A unified CM ->SIP trunk ->cube ->SIP trunk -> site B unified CM -> 6945(sip) | passed | |

Cisco Unified Contact Center Express

| Logical ID | Case Title | Description | Call Component Flow | Status | Defects |
|----------------------|--|--|---|--------|---------|
| UCCXJ861S CAD 013 | Call from SIP endpoint to Unified CCX Cisco Agent Desktop with Cisco Unified IP Phone 99XX via Unified Border Element | Verify the call from a SIP endpoint via Unified Border Element to Unified CCX Cisco Agent Desktop agent with Cisco Unified IP Phone 99XX works properly | Cisco Unified IP Phone A -> SIP ->Unified Border Element -> H.323 ->Unified CM -> Unified CCX ->Cisco Agent Desktop agent | Passed | |
| UCCXJ861S CAD 015 | Caller Entered Digit from SIP endpoint to Unified CCX Cisco Agent Desktop Agent Cisco Unified IP Phone 99XX via Unified Border Element | Verify the Caller Entered Digit from a SIP endpoint via Unified Border Element to Unified CCX Cisco Agent Desktop agent with Cisco Unified IP Phone 99XX works properly Configured as per RFC 2833 | Cisco Unified IP Phone A -> SIP ->Unified Border Element -> H.323 ->Unified CM -> Unified CCX ->Cisco Agent Desktop agent | Passed | |
| UCCXJ861S CAD 017 | Call from SIP endpoint to Unified CCX Cisco Agent Desktop Agent with Cisco Unified IP Phone 79xx via Unified Border Element | Verify the call from a SIP endpoint via Unified Border Element to Unified CCX Cisco Agent Desktop agent with SCCP Cisco Unified IP Phone 79xx works properly | Cisco Unified IP Phone A -> SIP ->Unified Border Element -> H.323 ->Unified CM -> Unified CCX ->Cisco Agent Desktop agent | Passed | |
| UCCXJ861S CAD 019 | Call from SIP endpoint to Unified CCX via Unified Border Element to Cisco Agent Desktop agent SIP Cisco Unified IP Phone 79xx | Verify the call from a SIP endpoint via Unified Border Element to Unified CCX Cisco Agent Desktop agent with SIP Cisco Unified IP Phone 79xx | Cisco Unified IP Phone A -> SIP ->Unified Border Element -> H.323 ->Unified CM -> Unified CCX ->Cisco Agent Desktop agent | Passed | |
| UCCXJ861S CBE 013 | Call from SIP endpoint to Unified CCX Cisco Agent Desktop—Browser Edition with Cisco Unified IP Phone 99XX via Unified Border Element | Verify the call from a SIP endpoint via Unified Border Element to Unified CCX Cisco Agent Desktop—Browser Edition withCisco Unified IP Phone 99XX works properly | Cisco Unified IP Phone A -> SIP ->Unified Border Element -> H.323 ->Unified CM -> Unified CCX ->Cisco Agent Desktop—Browser Edition | Passed | |

| Logical ID | Case Title | Description | Call Component Flow | Status | Defects |
|------------------------|--|--|---|--------|---------|
| UCCXJ861S CBE 015 | Caller Entered Digit from SIP endpoint to Unified CCX Cisco Agent Desktop—Browser Edition 99xx IP Phone via Unified Border Element | Verify the Caller Entered Digit from a SIP endpoint via Unified Border Element to Unified CCX Cisco Agent Desktop—Browser Edition with Cisco Unified IP Phone 99XX works properly Configured as per RFC 2833 | Cisco Unified IP Phone A -> SIP ->Unified Border Element -> H.323 ->Unified CM -> Unified CCX ->Cisco Agent Desktop—Browser Edition | Passed | |
| UCCXJ861S.CA D.001 | Inter-cluster call via Gatekeeper ICT to Unified CCX Cisco Agent Desktop with Cisco Unified IP Phone 9951 & 9971 | Verify inter-cluster basic call flow via Gatekeeper ICT to a Cisco Agent Desktop agent using Cisco Unified IP Phone 9951 & 9971 | Cisco Unified IP Phone A -> Unified CM -> GK ICT -> Unified CM -> Unified CCX -> Cisco Agent Desktop agent | Passed | |
| UCCXJ861 S.CAD .002 | Inter-cluster call via Non-Gatekeeper ICT to Unified CCX Cisco Agent Desktop agent with Cisco Unified IP Phone 9951 & 9971 | Verify inter-cluster basic call flow via Non-Gatekeeper ICT to a Cisco Agent Desktop agent using Cisco Unified IP Phone 9951 & 9971 | Cisco Unified IP Phone A -> Unified CM -> Non-GK ICT -> Unified CM -> Unified CCX -> Cisco Agent Desktop | Passed | |
| UCCXJ861 S.CAD .003 | Inter-cluster call via SIP Trunk to Unified CCX Cisco Agent Desktop agent with Cisco Unified IP Phone 9951 & 9971 | Verify inter-cluster basic call flow via SIP Trunk to a Cisco Agent Desktop agent using Cisco Unified IP Phone 9951 & 9971 | Cisco Unified IP Phone A -> Unified CM -> SIP -> Unified CM -> Unified CCX -> Cisco Agent Desktop agent | Passed | |
| UCCXJ861 S.CAD.021 | Cisco Agent Desktop Agent chat with Unified Personal Communicator user | Verify that the Cisco Desktop Agent agent is able to chat with the Unified Personal Communicator user from Cisco Agent Desktop agent. | | Passed | |
| UCCXJ861 S.CAD.022 | Cisco Agent Desktop Agent chat with Unified Personal Communicator user in remote site | Verify that the Cisco Agent Desktop agent is able to chat with the Unified Personal Communicator user in remote site | | Passed | |

| Logical ID | Case Title | Description | Call Component Flow | Status | Defects |
|-------------------------|---|--|--|--------|---------|
| UCCXJ861 S. CBE .003 | Inter-cluster call via SIP Trunk to Unified CCX Cisco Agent Desktop—Browser Edition with Cisco Unified IP Phone 9951 & 9971 | Verify inter-cluster basic call flow via SIP Trunk to a CAD BE agent using Cisco Unified IP Phone 9951& 9971 | Cisco Unified IP Phone A -> Unified CM -> SIP -> Unified CM -> Unified CCX -> Cisco Agent Desktop—Browser Edition | Passed | |
| UCCXJ861 S. CBE .004 | Inter-cluster call via Gatekeeper ICT to Unified CCX Cisco Agent Desktop—Browser Edition with 69xx Cisco Unified IP Phone | Verify inter-cluster basic call flow via gatekeeper ICT to a Cisco Agent Desktop—Browser Edition agent using 69xx Cisco Unified IP Phone | Cisco Unified IP Phone A -> Unified CM -> GK ICT -> Unified CM -> Unified CCX -> Cisco Agent Desktop—Browser Edition | Passed | |
| UCCXJ862S.CA D.33 | Verify the Cisco Agent Desktop agent login process | Verify the login process of Cisco Agent Desktop agent | | Passed | |
| UCCXJ862S.CA D.35 | Verify the login Cisco Agent Desktop agent through VPN | Verify the process of user can be login Cisco agent desktop by VPN | | Passed | |
| UCCXJ862S.CA D.36 | Retrieve the Agent call log display | Verify user can get back the Cisco Agent Desktop agent call logs display | | Passed | |
| UCCXJ862S.CS D.107 | Monitoring the Cisco Agent Desktop by Cisco supervisor desktop | Verify the process of Cisco Agent Desktop through Cisco supervisor desktop by the monitoring | | Passed | |
| UCCXJ862S.CS D.108 | Recording the conversation of Cisco Agent Desktop by Cisco supervisor desktop | verify whether the Cisco supervisor desktop is recording the conversation of Cisco agent desktop | | Passed | |
| UCCXJ862S.CS D.109 | Accessing the Agent call log records through Cisco supervisor desktop | Verify user can get back the Cisco Agent Desktop agent call log display from Cisco supervisor desktop | | Passed | |
| UCCXJ862S.HR. 103 | Initialized the Historical Reporting Client HRC Scheduler | Verify the user can initialized Historical Reporting Client scheduler | | Passed | |

Cisco Unified Presence

| Logical ID | Title | Description | Call Component Flow | Status | Defects |
|-------------------|---|--|---------------------|--------|---------|
| UCJ862S.CUP.G.101 | Unified Presence Upgrade | Verify that Unified Presence Upgrade to 8.6.3 successfully without any errors and intra/inter-cluster presence is functioning properly post-upgrade. | | Passed | |
| UCJ862S.CUP.G.102 | End users replication in Unified Presence server | Verify that users created in Cisco Unified Communications manager replicated in Cisco Unified Presence Server using LDAP sync up after upgrading Unified Presence to 8.6.3 | | Passed | |
| UCJ862S.CUP.G.104 | Presence status in Presence viewer page | Verify Presence Viewer displays correct buddy presence when contact list contains userid with space | | Passed | |
| UCJ862S.CUP.G.106 | Unified Presence authentication with the user having same DN in different partitions | Verify that Unified presence authentication status with the user having same DN in different partitions | | Passed | |
| UCJ862S.CUP.G.107 | Verifying Instant Message after upgrading Unified Presence to 8.6.3 | Verify whether Instant message delivered between two users successfully without any failures | | Passed | |
| UCJ862S.CUP.G.108 | Unified personal Communicator able to connect to Presence server when SIP proxy domain has space in it. | Verify whether Unified Personal Communicator able to connect to Presence server when SIP proxy domain has space in it. | | Passed | |
| UCJ862S.CUP.G.109 | Upgrading Unified Presence locale to latest build | Verify that latest locale build is upgraded to Unified Presence without any errors | | Passed | |
| UCJ862S.CUP.G.111 | Name change for Cisco Unified Personal Communicator in Unified presence GUI | Verify that Cisco unified personal communicator name changes to Cisco Jabber in all places of Unified Presence GUI as well as in online help | | Passed | |

| Logical ID | Title | Description | Call Component Flow | Status | Defects |
|-------------------|---|--|---------------------|--------|---------|
| UCJ862S.CUP.G.112 | Name change for Cisco Unified Personal Communicator in Unified Communications manager GUI | Verify that Cisco unified personal communicator name changes to Cisco Jabber in all places of unified Communications Manager GUI as well as in online help | | Passed | |
| UCJ862S.CUP.G.113 | Name change for desk phone control in Unified Presence GUI | Verify that desk phone control name changes to Microsoft RCC in all places of Unified Presence GUI as well as in online help | | Passed | |
| UCJ862S.CUP.G.114 | client types option in Unified Presence GUI | Verify that Client types option is available in GUI and user able to see all the default client type list | | Passed | |

Cisco Unity Connection

| Logical ID | Title | Description | Call Component Flow | Status | Defects |
|----------------|---|--|---------------------|--------|---------|
| UCJ862F.CUC.1 | Click Play and Pause while playing the voicemail in Cisco Personal Communications Assistant | Verify that the Cisco Personal Communications Assistant can play and pause the Voicemail | | Passed | |
| UCJ862F.CUC.2 | Click Play and Pause while playing the Voicemail in Visual Voicemail | Verify that the Visual Voicemail can play and pause the Voicemail | | Passed | |
| UCJ862F.CUC.3 | Delete the Voicemail Using Cisco Unified IP Phone 9951 and check same message is deleted in the exchange. | Verify that the Voicemail can be deleted in the exchange | | Passed | |
| UCJ862F.CUC.4 | Delete the Voicemail Using Cisco Unity ViewMail for Outlook (VMO) | Verify that the Voicemail can be deleted using Cisco Unity ViewMail for Outlook (VMO) | | Passed | |
| UCJ862F.CUC.5 | Forward a Voicemail with attachment using Cisco Personal Communications Assistant | Verify that the Voicemail can be forwarded to the destination | | Passed | |
| UCJ862F.CUC.6 | Forward a Voicemail Using Visual Voicemail | Verify that the Voicemail forwarded to the destination correctly | | Passed | |
| UCJ862F.CUC.7 | Increase or decrease the playback voice message speed using speed arrows in Cisco View mail for Outlook. | Verify that the playback increased or decreased using Speed Arrows | | Passed | |
| UCJ862F.CUC.8 | Mark read voicemail as unread voicemail in Microsoft outlook web access | Verify that the status has been changed to unread | | Passed | |
| UCJ862F.CUC.9 | New messages with MWI indicator Visual Voicemail | Verify that the Visual Voicemail showing the new message correctly | | Passed | |
| UCJ862F.CUC.10 | Playback Voicemail Recorded in Microsoft outlook web access | Verify that the Voicemail has been Played correctly in Outlook web access | | Passed | |
| UCJ862F.CUC.11 | Playback Voicemail Recorded in Cisco ViewMail for Outlook | Verify that the Voicemail has been Played correctly in Cisco Viewmail for Outlook | | Passed | |

| Logical ID | Title | Description | Call Component Flow | Status | Defects |
|----------------|---|--|---|--------|---------|
| UCJ862F.CUC.12 | Progress and Length of the message recorded during play back in Visual Voicemail | Verify that the Progress and Length of the message is shown | | Passed | |
| UCJ862F.CUC.13 | Record & Send a Voicemail Using Visual Voicemail | Verify that the Voicemail is recorded using Visual Voicemail | | Passed | |
| UCJ862F.CUC.14 | Record & Send a Voicemail using Unified IP Phone to Cisco Cius | Verify that the Voicemail is recorded and sent to the destination | Unified IP Phone-->Unified CM-->Cisco Cius | Passed | |
| UCJ862F.CUC.15 | Reply voicemail with attachment in Visual Voicemail | Verify that the Voicemail has been reached to the destination | | Passed | |
| UCJ862F.CUC.16 | Send voicemail to single inbox enabled user using Cisco Personal Communications Assistant | Verify that the Voicemail has been reached to the destination | | Passed | |
| UCJ862F.CUC.17 | Send a Voicemail to the user of Cisco Unity ViewMail for Outlook | Verify that the user can able to check the voicemail in Viewmail | | Passed | |
| UCJ862F.CUC.18 | Volume bar adjust while playing a voicemail in Visual Voicemail | Verify that the User can adjust the Volume bar while Playing the voicemail | | Passed | |
| UCJ862F.CUC.19 | Call from Unified IP Phone 9951 via H.323 trunk to phone with CFNA to Cisco Cisco Cius phone and connects to Voicemail | Verify that the Unified IP Phone 9951 sends voicemail to the Cisco Cius | Unified IP Phone 9951->Unified CM-->H.323-->Unified CM-->Cisco Cius-->CUC | Passed | |
| UCJ862F.CUC.20 | Record and send a voice mail from Unified IP Phone 9951 to Unified IP Phone 7925 and Retrieve the voice mail from Unified IP Phone 7925 | Verify that the Cisco Unity Connection handles the Voicemail correctly and delivered it to the Correct destination | Unified IP Phone 7921->Unified CM->Unified IP Phone 7925->CUC | Passed | |
| UCJ862F.CUC.21 | Record and send a voice mail from Unified IP Phone 8945 to Unified IP Phone 9971 and Retrieve the voice mail from 9971 | Verify that the Cisco Unity Connection handles the Voicemail correctly and delivered it to the Correct destination | Unified IP Phone 8945->Unified CM->Unified IP Phone 9971->CUC | Passed | |
| UCJ862F.CUC.22 | Record and send a voice mail from Unified IP Phone 8941 to Cisco Cius and Retrieve the voice mail from Cisco Cius | Verify that the Cisco Unity Connection handles the Voicemail correctly and delivered it to the Correct destination | Unified IP Phone 8941->Unified CM->Cisco Cius->CUC | Passed | |

| Logical ID | Title | Description | Call Component Flow | Status | Defects |
|----------------|---|--|---|--------|---------|
| UCJ862F.CUC.23 | Send voice message marking private using Cisco Personal Communications Assistant | Verify that the Voicemail has been sent when it was marked as Private | | Passed | |
| UCJ862F.CUC.24 | Send voice message marking urgent in Cisco Personal Communications Assistant | Verify that the Voicemail has been sent when it was marked as Urgent | | Passed | |
| UCJ862F.CUC.25 | Upload .wav format recorded voicemail and send using Cisco Personal Communications Assistant | Verify that the voicemail has been uploaded to sent using Cisco Personal Communications Assistant | | Passed | |
| UCJ862F.CUC.26 | Call from Unified IP Phone 9951 SIP phone via H.323 trunk to phone with CFNA to Cisco Unified IP Phone 8941 to Voicemail | Verify that the Cisco Unity Connection handles the Voicemail correctly and delivered it to the Correct destination | Unified IP Phone 9951-->Unified CM-->H.323-->Unified IP Phone 8941->CUC | Passed | |
| UCJ862F.CUC.27 | Call from Unified IP Phone 9971 SIP phone to Unified IP Phone 9951 with Call Forward No Answer to Cisco IP Video Phone E20 to Voicemail | Verify that the Cisco Unity Connection handles the Voicemail correctly and delivered it to the Correct destination | Unified IP Phone 9971-->Unified CM-->Unified IP Phone 9951-->CFNA-->Cisco IP Video Phone E20->CUC | Passed | |
| UCJ862F.CUC.28 | Call from Unified IP Phone 8945 via H.323 trunk - Call Forward All configured in Cisco IP Video Phone E20 to Unified IP Phone 9951 with CFNA to Voicemail | Verify that the Cisco Unity Connection handles the Voicemail correctly and delivered it to the Correct destination | Cisco Unified IP Phone 8945->Unified CM->H.323->Unified Video IP Phone E20->CFALL-->Unified IP Phone 9951 CFNA to Voicemail | Passed | |
| UCJ862F.CUC.29 | Call from Cisco TelePresenceSystem EX60 via H.323 trunk - Call Forward All from Unified IP Phone 8941 to Unified IP Phone 9951 with CFNA to Voicemail | Verify that the Cisco Unity Connection handles the Voicemail correctly and delivered it to the Correct destination | Cisco TelePresenceSystem EX60->Unified CM->H.323->Unified CM-->Unified IP Phone 8941->CFALL-->Unified IP Phone 9951 CFNA to Voicemail | Passed | |

| Logical ID | Title | Description | Call Component Flow | Status | Defects |
|----------------|--|--|---|--------|---------|
| UCJ862F.CUC.30 | Call from Cisco TelePresenceSystem EX90 via H.323 trunk - Call Forward All from Cisco IP Video Phone E20 to Unified IP Phone 9951 SIP with CFNA to Voicemail | Verify that the Cisco Unity Connection handles the Voicemail correctly and delivered it to the Correct destination | Cisco TelePresenceSystem EX90->Unified CM->H.323->Unified CM-->E20->CFALL-->Unified IP Phone 9951 CFNA to Voicemail | Passed | |
| UCJ862F.CUC.31 | Record and send a voice mail from Cisco Video IP Phone E20 to Cisco Unified IP Phone 8941 and Retrieve the voice mail from Unified IP Phone 8941. | Verify that the Cisco Unity Connection handles the Voicemail correctly and delivered it to the Correct destination | Unified IP Phone E20->Unified CM->CUC->Unified IP Phone 8941-->CUC | Passed | |
| UCJ862F.CUC.32 | Record and send a voice mail from Unified IP Phone 8941 to Unified IP Phone 9971 and Retrieve the voice mail from Unified IP Phone 9971 | Verify that the Cisco Unity Connection handles the Voicemail correctly and delivered it to the Correct destination | Unified IP Phone 8941>Unified CM->CUC->Unified IP Phone 9971-->CUC | Passed | |
| UCJ862F.CUC.33 | Record and send a voice mail from Unified IP Phone 8945 to Cisco Video IP Phone E20 and Retrieve the voice mail from Cisco Video IP Phone E20 | Verify that the Cisco Unity Connection handles the Voicemail correctly and delivered it to the Correct destination | Unified IP Phone 8945->Unified CM->Unified IP Phone E20-->CUC | Passed | |

Cisco Unified Personal Communicator

| Logical ID | Title | Description | Call Component Flow | Status | Defects |
|------------------------|--|--|--|--------|---------|
| UCJ862S.CUP C.G.101 | Login to Unified Personal Communicator after upgrade | Verify that user able to login to Unified Personal Communicator with his profile successfully | | Passed | |
| UCJ862S.CUP C.G.103 | Unified Personal Communicator with video call (soft phone mode) | Verify that Unified Personal Communicator as soft phone initiates video call with another video Unified IP Phone and establish connection without any errors | | Passed | |
| UCJ862S.CUP C.G.105 | Unified Personal Communicator with video call (Desk phone mode) | Verify that Unified Personal Communicator as desk phone initiates video call with another video IP Phone and establish connection without any errors | | Passed | |
| UCJ862S.CUP C.G.106 | Unified Personal Communicator Music On Hold | Verify that user able to listen Music on hold in Unified Personal Communicator | | Passed | |
| UCJ862S.CUP C.G.107 | Unified Personal Communicator registration status when PC unplugged from network | Verify the status of Unified Personal Communicator when unplug the network cable from PC and plug the network cable | | Passed | |
| UCJ862S.CUP C.G.111 | Video call between Unified Personal Communicator as soft phone and E20 | Verify that video call happens between Unified Personal Communicator as soft phone and E20 | Unified Personal Communicator ->Unified Communications manager->Site A->(SIP Trunk)->Site B-> Unified Communications manager ->E20 | Passed | |

| Logical ID | Title | Description | Call Component Flow | Status | Defects |
|------------------------|---|--|---|--------|---------|
| UCJ862S.CUP C.G.112 | Video call between Unified Personal Communicator as soft phone and EX60 | Verify that video call happens between Unified Personal Communicator as soft phone and EX60 | Unified Personal Communicator -> Unified Communications manager ->Site A->(SIP Trunk)->Site B-> Unified Communications manager ->EX60 | Passed | |
| UCJ862S.CUP C.G.114 | Video call between Unified Personal Communicator as desk phone mode and E20 | Verify that video call happens between Unified Personal Communicator as desk phone mode and E20 | Unified Personal Communicator -> Unified Communications manager Site A->(SIP Trunk)->Site B-> Unified Communications manager ->E20 | Passed | |
| UCJ862S.CUP C.G.115 | Video call between Unified Personal Communicator as desk phone mode and EX60 | Verify that video call happens between Unified Personal Communicator as desk phone mode and EX60 | Unified Personal Communicator -> Unified Communications manager ->Site A->(SIP Trunk)->Site B-> Unified Communications manager -Ex60 | Passed | |
| UCJ862S.CUP C.G.117 | Forward All feature between Unified Personal Communicator as soft phone and E20 | Verify that all calls coming from Unified Personal Communicator as soft phone to E20 forwarding to EX60/90 | Unified Personal Communicator -> Unified Communications manager ->Site A->(SIP Trunk)->Site B-> Unified Communications manager ->E20(CFwdAll)->EX60/90 | Passed | |

| Logical ID | Title | Description | Call Component Flow | Status | Defects |
|------------------------|---|---|--|--------|---------|
| UCJ862S.CUP C.G.118 | Forward All feature between Unified Personal Communicator as soft phone and EX60 | Verify that all calls coming from Unified Personal Communicator as soft phone to EX60 forwarding to E20 | Unified Personal Communicator -> Unified Communication s manager ->Site A->(SIP Trunk)->Site B-> Unified Communication s manager ->EX60(CFwd All)->E20 | Passed | |
| UCJ862S.CUP C.G.121 | Forward All feature between Unified Personal Communicator as desk phone mode and EX60 | Verify that all calls coming from Unified Personal Communicator as desk phone mode to EX60 forwarding to E20 | Unified Personal Communicator -> Unified Communication s manager ->Site A->(SIP Trunk)->Site B-> Unified Communication s manager ->EX60(CFwd All)->E20 | Passed | |
| UCJ862S.CUP C.G.126 | Transfer feature between Unified Personal Communicator as Desk phone mode and E20 | Verify that call coming from Unified Personal Communicator as desk phone mode to E20 able to transfer EX60/90 | Unified Personal Communicator -> Unified Communication s manager ->Site A->(SIP Trunk)->Site B-> Unified Communication s manager ->E20(Tranfer)->EX6090 | Passed | |

| Logical ID | Title | Description | Call Component Flow | Status | Defects |
|------------------------|--|--|---|--------|---------|
| UCJ862S.CUP C.G.127 | Transfer feature between Unified Personal Communicator as desk phone mode and EX60 | Verify that call coming from Unified Personal Communicator as desk phone mode to EX60 able to transfer E20 | Unified Personal Communicator -> Unified Communications manager ->Site A->(SIP Trunk)->Site B-> Unified Communications manager ->EX60(Transfer)->E20 | Passed | |
| UCJ862S.CUP C.G.129 | Instant message between two Unified Personal Communicators | Verify that Instant message sent and received without any errors between two unified Personal communicators | Unified Personal Communicator 1->unified CM->Site A->SIP trunk->Site B->unified CM-> Unified Personal Communicator 2 | Passed | |
| UCJ862S.CUP C.G.130 | Instant message between Unified Personal Communicator and Cisco Jabber IM | Verify that Instant message sent and received without any errors between unified Personal communicator and Cisco Jabber IM | Unified Personal Communicator 1->unified CM->Site A->SIP trunk->Site B->unified CM-> Cisco Jabber IM | Passed | |

Cisco UC Integration™ for Microsoft LYNC

| Logical ID | Title | Description | Call Component Flow | Status | Defects |
|-------------------|-------------------------------|--|---|--------|---------|
| UCJ862F.CL.U.010 | CUCI-Lync client installation | Verify that the installation of the client is successful | | Passed | |
| UCJ862F.CL.U.011 | CUCI-Lync- Call notification | Verify that there is an incoming call notification on the client | Unified IP Phone A->Unified CM 1->Client 1 | Passed | |
| UCJ862F.CLR.U.002 | Presence status | Verify that the presence status of the user is shown correctly | | Passed | |
| UCJ862F.CL.U.002 | Intercluster call | Verify that an intercluster call is successful | Client 1->Unified CM 1->ICT->Unified CM 2->Unified IP Phone A | Passed | |
| UCJ862F.CL.U.006 | Deskphone startup mode | Verify that Client works in Desktop mode | Unified IP Phone A->Unified CM 1->Client 1 (shares the same DN as Unified IP Phone B) | Passed | |
| UCJ862F.CL.U.009 | Account lookup prevention | Verify that Account lookup prevention works successful | | Passed | |
| UCJ862F.CL.U.004 | Forced Authorization codes | Verify that Forced Authorization Code works successful | Unified IP Phone A->Unified CM 1->SIP->Unified CM 2-> Client 1 | Passed | |
| UCJ862F.CL.U.008 | Ringtone Customization | Verify that ringtone settings can be customized in the client | Unified IP Phone A->Unified CM 1->Client 1 | Passed | |
| UCJ862F.CLR.U.001 | Call Park | Verify that call park is successful | | Passed | |
| UCJ862F.CL.U.007 | Instant Messaging | Verify that chat between two clients are successful | | Passed | |

Cisco Unified Personal Communicator on Mac

| Logical ID | Title | Description | Call Component Flow | Status | Defects |
|---------------------------------|--|--|---------------------|--------|---------|
| UCJ862S.CUP ConMAC.G.10 1 | Login to Unified personal Communicator installed on MAC Operating System after upgrade | Verify that user able to login to Unified personal Communicator installed on MAC Operating System with his profile successfully | | Passed | |
| UCJ862S.CUP ConMAC.G.10 3 | Unified personal Communicator installed on MAC Operating System with video call (soft phone mode) | Verify that Unified personal Communicator installed on MAC Operating System as soft phone initiates video call with another video Unified IP Phone and establish connection without any errors | | Passed | |
| UCJ862S.CUP ConMAC.G.10 5 | Unified personal Communicator installed on MAC Operating System with video call (Desk phone mode) | Verify that Unified personal Communicator installed on MAC Operating System as desk phone initiates video call with another video IP Phone and establish connection without any errors | | Passed | |
| UCJ862S.CUP ConMAC.G.10 6 | Unified personal Communicator installed on MAC Operating System Music On Hold | Verify that user able to listen Music on hold in Unified personal Communicator installed on MAC Operating System | | Passed | |
| UCJ862S.CUP ConMAC.G.10 7 | Unified personal Communicator installed on MAC Operating System registration status when PC unplugged from network | Verify the status of Unified personal Communicator installed on MAC Operating System when unplug the network cable from PC and plug the network cable | | Passed | |

| Logical ID | Title | Description | Call Component Flow | Status | Defects |
|---------------------------------|---|---|---|--------|---------|
| UCJ862S.CUP ConMAC.G.11 1 | Video call between Unified personal Communicator installed on MAC Operating System as soft phone and E20 | Verify that video call happens between Unified personal Communicator installed on MAC Operating System as soft phone and E20 | Unified personal Communicator installed on MAC Operating System ->Unified Communication s manager->Site A->(SIP Trunk)->Site B-> Unified Communication s manager ->E20 | Passed | |
| UCJ862S.CUP ConMAC.G.11 2 | Video call between Unified personal Communicator installed on MAC Operating System as soft phone and EX60 | Verify that video call happens between Unified personal Communicator installed on MAC Operating System as soft phone and EX60 | Unified personal Communicator installed on MAC Operating System -> Unified Communication s manager ->Site A->(SIP Trunk)->Site B-> Unified Communication s manager ->EX60 | Passed | |
| UCJ862S.CUP ConMAC.G.11 4 | Video call between Unified personal Communicator installed on MAC Operating System as desk phone mode and E20 | Verify that video call happens between Unified personal Communicator installed on MAC Operating System as desk phone mode and E20 | Unified personal Communicator installed on MAC Operating System -> Unified Communication s manager Site A->(SIP Trunk)->Site B-> Unified Communication s manager ->E20 | Passed | |

| Logical ID | Title | Description | Call Component Flow | Status | Defects |
|---------------------------------|--|--|--|--------|---------|
| UCJ862S.CUP ConMAC.G.11 5 | Video call between Unified personal Communicator installed on MAC Operating System as desk phone mode and EX60 | Verify that video call happens between Unified personal Communicator installed on MAC Operating System as desk phone mode and EX60 | Unified personal Communicator installed on MAC Operating System -> Unified Communications manager ->Site A->(SIP Trunk)->Site B-> Unified Communications manager -Ex60 | Passed | |
| UCJ862S.CUP ConMAC.G.11 7 | Forward All feature between Unified personal Communicator installed on MAC Operating System as soft phone and E20 | Verify that all calls coming from Unified personal Communicator installed on MAC Operating System as soft phone to E20 forwarding to EX60/90 | Unified personal Communicator installed on MAC Operating System -> Unified Communications manager ->Site A->(SIP Trunk)->Site B-> Unified Communications manager ->E20(CFwdAll)->EX60/90 | Passed | |
| UCJ862S.CUP ConMAC.G.11 8 | Forward All feature between Unified personal Communicator installed on MAC Operating System as soft phone and EX60 | Verify that all calls coming from Unified personal Communicator installed on MAC Operating System as soft phone to EX60 forwarding to E20 | Unified personal Communicator installed on MAC Operating System -> Unified Communications manager ->Site A->(SIP Trunk)->Site B-> Unified Communications manager ->EX60(CFwdAll)->E20 | Passed | |

| Logical ID | Title | Description | Call Component Flow | Status | Defects |
|---------------------------------|---|--|--|--------|---------|
| UCJ862S.CUP ConMAC.G.12 1 | Forward All feature between Unified personal Communicator installed on MAC Operating System as desk phone mode and EX60 | Verify that all calls coming from Unified personal Communicator installed on MAC Operating System as desk phone mode to EX60 forwarding to E20 | Unified personal Communicator installed on MAC Operating System -> Unified Communication s manager ->Site A->(SIP Trunk)->Site B-> Unified Communication s manager ->EX60(CFwd All)->E20 | Passed | |
| UCJ862S.CUP ConMAC.G.12 3 | Transfer feature between Unified personal Communicator installed on MAC Operating System as soft phone and E20 | Verify that call coming from Unified personal Communicator installed on MAC Operating System as soft phone to E20 able to transfer EX60/90 | Unified personal Communicator installed on MAC Operating System -> Unified Communication s manager ->Site A->(SIP Trunk)->Site B-> Unified Communication s manager ->E20(Tranfer)->EX60/90 | Passed | |
| UCJ862S.CUP ConMAC.G.12 4 | Transfer feature between Unified personal Communicator installed on MAC Operating System as soft phone and EX60 | Verify that call coming from Unified personal Communicator installed on MAC Operating System as soft phone to EX60 able to transfer E20 | Unified personal Communicator installed on MAC Operating System -> Unified Communication s manager ->Site A->(SIP Trunk)->Site B-> Unified Communication s manager ->EX60(Tranfer)->E20 | Passed | |

| Logical ID | Title | Description | Call Component Flow | Status | Defects |
|---------------------------------|--|---|---|--------|---------|
| UCJ862S.CUP ConMAC.G.12 6 | Transfer feature between Unified personal Communicator installed on MAC Operating System as Desk phone mode and E20 | Verify that call coming from Unified personal Communicator installed on MAC Operating System as desk phone mode to E20 able to transfer EX60/90 | Unified personal Communicator installed on MAC Operating System -> Unified Communication s manager ->Site A->(SIP Trunk)->Site B-> Unified Communication s manager ->E20(Tranfer)->EX6090 | Passed | |
| UCJ862S.CUP ConMAC.G.12 7 | Transfer feature between Unified personal Communicator installed on MAC Operating System as desk phone mode and EX60 | Verify that call coming from Unified personal Communicator installed on MAC Operating System as desk phone mode to EX60 able to transfer E20 | Unified personal Communicator installed on MAC Operating System -> Unified Communication s manager ->Site A->(SIP Trunk)->Site B-> Unified Communication s manager ->EX60(Tranfer)->E20 | Passed | |
| UCJ862S.CUP ConMAC.G.12 8 | Transfer feature between Unified personal Communicator installed on MAC Operating System as desk phone mode and EX90 | Verify that call coming from Unified personal Communicator installed on MAC Operating System as desk phone mode to EX90 able to transfer E20 | Unified personal Communicator installed on MAC Operating System -> Unified Communication s manager ->Site A->(SIP Trunk)->Site B-> Unified Communication s manager ->EX90(Tranfer)->E20 | Passed | |

| Logical ID | Title | Description | Call Component Flow | Status | Defects |
|---------------------------------|---|--|--|--------|---------|
| UCJ862S.CUP ConMAC.G.12 9 | Instant message between two Unified personal Communicator installed on MAC Operating Systems | Verify that Instant message sent and received without any errors between two Unified personal Communicator installed on MAC Operating Systems | Unified personal Communicator installed on MAC Operating System 1->unified CM->Site A->SIP trunk->Site B->unified CM-> Unified personal Communicator installed on MAC Operating System 2 | Passed | |
| UCJ862S.CUP ConMAC.G.13 0 | Instant message between Unified personal Communicator installed on MAC Operating System and Cisco Jabber IM | Verify that Instant message sent and received without any errors between Unified personal Communicator installed on MAC Operating System and Cisco Jabber IM | Unified personal Communicator installed on MAC Operating System 1->unified CM->Site A->SIP trunk->Site B->unified CM-> Cisco Jabber IM | Passed | |

Cisco Unified Communications Manager Express

| Logical ID | Title | Description | Call Component Flow | Status | Defects |
|------------------|---|---|---|--------|---------|
| UC862S.CME.U.015 | Call Transfer between Phones registered to Cisco Unified Communication Manager and Cisco Unified Communication Manager Express | Verify whether able to attempt Call Transfer between Phones registered to Unified CM and Unified CME Works Successfully | IP Phone 8961->Unified CME-> Unified Border Element->Cisco Unified CM->IP Phone 9951->Call Transfer Cisco Unified IP Phone 8961->Cisco Unified CME->IP Phone 8941 | Passed | |
| UC862S.CME.U.016 | Call Forward between Phones Registered to Cisco Unified Communication Manager and Cisco Unified Communication Manager Express | Verify that Call Forward between Phones registered to Unified CM and Unified CME works Successfully | IP Phone 7985G->Unifield CME->Unified Border Element->Cisco Unified CM->IP Phone 8941->Call Forward IP Phone 7985G->Unifield CME->IP Phone 7911G | Passed | |
| UC862S.CME.U.021 | Hold and Resume in Calls between Phones registered to Cisco Unified Communication Manager and Cisco Unified Communication Manager Express | Verify that Hold and Resume in Calls between Phones registered to Unified CM and Unified CME works Successfully | IP Phone 8945(Phone A)->Unified CME->Unifed Border Element->Unifield CM->IP Phone 9951(Phone B)->Hold IP Phone 8945(Phone A)->Resume | Passed | |

| Logical ID | Title | Description | Call Component Flow | Status | Defects |
|----------------------|--|---|--|--------|---------|
| UC862S.CME. U.006 | Mixed Shared Line in Cisco Unified Communication Manager Express | Verify That Mixed Shared Line Works Successfully | IP Phone 6921(SCCP Phone)-> Unified CME->IP Phone 9951(SIP Phone) | Passed | |
| UC862S.CME. U.009 | Hold and Resume on Mixed Shared Line | Verify that Hold and Resuming a Call On Mixed Shared Line is Successful | IP Phone 9971->Unified CME->IP Phone 6921->Hold IP Phone 6921->Resume | passed | |
| UC862S.CME. U.010 | Call Forward on Mixed Shared Line | Verify That Call Forward On Mixed Shared Line Works Successfully | IP Phone 9951->Unified CME->IP Phone 9971->Call Forward IP Phone 9951->Unified CME->IP Phone 6921 | Passed | |
| UC862S.CME. U.011 | Blind Transfer on Mixed Shared Line | Verify that Blind Transfer On Mixed Shared Line works Successfully | IP Phone 6921-> Unified CME->IP Phone 6941->Blind Transfer Cisco Unified IP Phone 6921->Unified CME->IP Phone 9951 | Passed | |
| UC862S.CME. U.012 | Consult Transfer On Mixed Shared Line | Verify that Consult Transfer On Mixed Shared Line works Successfully | IP Phone 6921->Unified CME->IP Phone 6941-> Consult Transfer IP Phone 6921->Unified CME-> IP Phone 9951 | passed | |

| Logical ID | Title | Description | Call Component Flow | Status | Defects |
|------------------|--|--|---|--------|---------|
| UC862S.CME.U.013 | Multiple Calls Per Line (MCPL) Feature on Cisco Unified IP Phone 6921,6941(SIP) In Cisco Unified Communication Manager Express 9.0 | Verify that Multiple Calls Per Line works successfully | IP Phone 6921->Unified CME->Multiple CallsPerLine IP Phone 6921->Unified CME->IP Phone 6941,6961 | Passed | |
| UC862S.CME.U.030 | Hardware Conferencing In Cisco Unified IP Phone 9951 | Verify whether able to attempt Hardware Conferencing and if it is successful. | IP Phone 9951->Unified CME->IP Phone 9971->Confere nce->IP Phone 8961,IP Phone 8941 IP Phone 8961->Join IP Phone 8941->Join | passed | |
| UC862S.CME.U.002 | Call Blocking Based on Date and Time (After-Hours Toll Bar) In Cisco Unified IP Phone 6921. | Verify if a call can be blocked between Phones registered to Unified CME and Unified CM. | 6921 Phone->Unified CME (After Hours) | Passed | |
| UC862S.CME.U.032 | Single Number Reach In Cisco Unified IP Phone 9951 | Verify Whether Single Number Reach Works Successfully In Cisco Unified IP Phone 9951 | IP Phone 6945->Unified CME->IP Phone 9951(Unanattended)->Mobility IP Phone 6945->Unified CME->Cisco Unified Border Element->Unifi ed CM->IP Phone 9971 | Passed | |

| Logical ID | Title | Description | Call Component Flow | Status | Defects |
|----------------------|---------------------------------|--|---|--------|---------|
| UC862S.CME. U.033 | Hold and Resume in ATA-187 | Verify that Hold and Resuming a call in ATA-187 works successfully | Analog Phone->ATA-1 87->Cisco Unified Communication Manager Express->Cisco Unified IP Phone 8941(Phone B)->Hold Analog Phone->ATA-1 87->Resume | passed | |
| UC862S.CME. U.034 | Attended Transfer in ATA-187 | Verify that Attended Transfer Works Successfully In ATA-187 | Analog Phone->ATA-1 87->Unified CME->IP Phone 8941(Phone B)->Hold->Tra nsfer Analog Phone->ATA-1 87->Unified CME->IP Phone 9951(Dialed Party) | Passed | |
| UC862S.CME. U.035 | Call Pick Up in ATA-187 | Verify that Call Pick Up works successfully In ATA-187 | Analog Phone->ATA-1 87->Unified CME->IP Phone 8941(Phone B)->Call Pick Up Analog Phone->ATA-1 87->Unified CME->IP Phone 9951(User In the Pick-Up Group) | passed | |

| Logical ID | Title | Description | Call Component Flow | Status | Defects |
|----------------------|--|---|--|--------|---------|
| UC862S.CME. U.039 | Meet-Me Conference in ATA-187 | Verify that Meet-Me Conference in ATA-187 works successfully | Analog Phone->ATA-187->Unified CME->IP Phone 8941(Phone B)->Meet-Me Conference Number | Passed | |
| UC862S.CME. U.040 | Shared Line in ATA-187 | Verify that Shared Line in ATA-187 works successfully | Analog Phone->ATA-187->Cisco Unified Communication Manager Express->Cisco Unified IP Phone 8941(Phone B) (Same Directory Number) | passed | |
| UC862S.CME. U.061 | Call Park On Mixed Shared Line | Verify that Call Park Works successfully in Mixed Shared Line works successfully | IP Phone 9951->Unified CME-> IP Phone 6961->Park | Passed | |
| UC862S.CME. U.072 | Meet-Me Conference in Cisco Unified IP Phone 7911G | Verify that Meet-Me Conference in Cisco Unified IP Phone 7911G works successfully | IP Phone 7911G->Unifield CME->Meet-Me Conference Number IP Phone 7921G->Meet-Me Conference Number IP Phone 8941->Meet-Me Conference Number | passed | |

| Logical ID | Title | Description | Call Component Flow | Status | Defects |
|----------------------|---|---|--|--------|---------|
| UC862S.CME. U.073 | Meet-Me Conference in Cisco Unified Wireless IP Phone 7921G | Verify That Phone user can host a Meet-Me conference in which the participants call a predetermined number at a scheduled time. | IP Phone 7921G->Unified CME->Meet-Me Conference Number IP Phone 8945->Meet-Me Conference Number IP Phone 6945->Meet-Me Conference Number | Passed | |
| UC862S.CME. U.074 | Shared Line in Cisco Unified IP Phone 7911G | Verify that Phone user Connected can Share a line with another Phone with the Same Directory Number. | IP Phone 7911G->Unified CME->IP Phone 8941(Phone B) (Same Directory Number) | passed | |
| UC862S.CME. U.075 | Shared Line in Cisco Unified Wireless IP Phone 7921G | Verify that Shared Line in Unified IP Phone 7921G successfully | IP Phone 7921G->Unified CME->Unified IP Phone 8941(Phone B) (Same Directory Number) | Passed | |
| UC862S.CME. U.076 | Speed Dial in Cisco Wireless IP Phone 7921G | Verify that Speed Dial In Unified IP Phone 7921G works successfully | IP Phone 7921G ->Unified CME->IP Phone 7925G ->Speed Dial | Passed | |
| UC862S.CME. U.077 | Call Park In Cisco Unified Wireless IP Phone 7921G. | Verify that Call Park works successfully in Unified IP Phone 7921G. | IP Phone 7921G->Unified CME->IP Phone 7925G IP Phone 7921G->Park | Passed | |

| Logical ID | Title | Description | Call Component Flow | Status | Defects |
|------------------|---|---|---|--------|---------|
| UC862S.CME.U.078 | Call Park In Cisco Unified Wireless IP Phone 7925G. | Verify that Call Park works successfully in Unified IP Phone 7925G. | IP Phone 7925G->Unified CME->IP Phone 7921G IP Phone 7925G->Park | Passed | |
| UC862S.CME.U.079 | Call Pick Up in Cisco Unified Wireless IP Phone 7921G | Verify that Call Pick Up works successfully In Unified IP Phone 7921G | IP Phone 7921G->Unified CME->Unified IP Phone 7911G(Phone B)->Call Pick Up IP Phone 7921G->Unified CME->IP Phone 7926G(User In the Pick-Up Group) | Passed | |
| UC862S.CME.U.080 | Call Pick Up in Cisco Unified Wireless IP Phone 7925G | Verify that Call Pick Up works successfully In Unified IP Phone 7925G | IP Phone 7925G->Unified CME->IP Phone 7911G(Phone B)->Call Pick Up IP Phone 7925G->Unified CME->IP Phone 7921G(User In the Pick-Up Group) | Passed | |
| UC862S.CME.U.082 | Hold and Resume in Cisco Unified Wireless Phone 7921G | Verify that Hold and Resuming a call works successfully in Unified IP Phone 7921G | IP Phone 7921G->Unified CME->IP Phone 7925G->Hold IP Phone 7925G->Resume | Passed | |

| Logical ID | Title | Description | Call Component Flow | Status | Defects |
|----------------------|--|--|---|--------|---------|
| UC862S.CME. U.084 | Call Transfer in Cisco Unified Wireless IP Phone 7921G | Verify whether able to attempt Call Transfer in Unified IP Phone 7921G and transfer is successful. | IP Phone 7921G-> Unified CME->IP Phone 7925G->Call Transfer IP Phone 7921G->Unified CME->IP Phone 7985G | Passed | |
| UC862S.CME. U.085 | Call Forward All in Cisco Unified IP Phone 7921G | Verify that CallFwd All Works Successfully between Two Sites | IP Phone 8945->Unified CME->IP Phone 7921G->Call Forward All IP Phone 8945->Cisco Unified CME->IP Phone 7911G | Passed | |
| UC862S.CME. U.086 | Call Forward Between Two Sites | Verify that Call Forward works successful between Two Sites | IP Phone 9951->Unified CME-> Unified Border Element->SIP Unified CME->Cisco Unified IP Phone 8961->Call Forward IP Phone 9951->Unified CME->SIP IP Phone 3905 | Passed | |

| Logical ID | Title | Description | Call Component Flow | Status | Defects |
|----------------------|--|---|---|--------|---------|
| UC862S.CME. U.087 | Call Transfer Between Phones Registered to SIP Cisco Unified Communication Manager Express and Cisco Unified Communication Manager Express in Site A | Verify that Call Transfer Between Phones Registered to SIP Unified CME and Unified CME works successfully | IP Phone 8961->SIP Unified CME->Unified Border Element->Unified CME->IP Phone 9971->Call Transfer IP Phone 8961->SIP Unified CME > Unified Border Element->Unified CME-> IP Phone 7921G | passed | |
| UC862S.CME. U.088 | Hold and Resume Between Phones Registered to SIP Cisco Unified Communication Manager Express Site and Site A Cisco Unified Communication Manager Express | Verify That Hold and Resume Between Phones Registered to SIP Unified CME Site and Unified CME works Successfully | IP Phone 8961->SIP Unified CME->Unified Border Element->Unified CME->IP Phone 7985->Hold IP Phone 8961->Resume | Passed | |
| UC862S.CME. U.054 | Extension Mobility in Cisco Unified IP Phone 6961 | Verify if the phone user can make and receive calls on Extension Mobility Enabled phone using the same personal directory number as is on their own desk phone. | IP Phone 6961->Extension Mobility EM IP Phone 6961->Unified CME ->IP Phone8945 EM IP Phone 6961->Logout | passed | |
| UC862S.CME. U.004 | Extension Mobility in Cisco Unified IP Phone 6921 | Verify if the phone user can make and receive calls on Extension Mobility Enabled phone using the same personal directory number as is on their own desk phone. | IP Phone 6921->Extension Mobility EM IP Phone 6921->Unified CME ->IP Phone8945 EM IP Phone 6921->Logout | passed | |

| Logical ID | Title | Description | Call Component Flow | Status | Defects |
|----------------------|---|--|--|--------|---------|
| UC862S.CME. U.093 | UNIFIED CME Video Conferencing- Transfer | Verify that transfer feature is working in video Conferencing | 9951-> UNIFIED CME-> 9951-> 9971->Transfer | passed | |
| UC862S.CME. U.094 | UNIFIED CME Video Conferencing- Call Forward | Verify that Call forward Feature is working in Video Conferencing | 9951-> UNIFIED CME-> 9951-> 9971->Call Forward->9951 | passed | |
| UC862S.CME. U.095 | UNIFIED CME Video Conferencing- Call Pickup | Verify that Call pickup Feature is working in video conferencing | 9951-> UNIFIED CME-> 9951-> 9971->Call Pickup | passed | |
| UC862S.CME. U.097 | UNIFIED CME Video Conferencing- Hold | Verify that Hold feature is working in Homogeneous Video Conferencing | 9951-> UNIFIED CME-> 9951-> 9971->Hold | passed | |
| UC862S.CME. U.098 | UNIFIED CME Video Conferencing - Heterogeneous | Verify that Hold feature is working in Heterogeneous Video Conferencing | 9951-> UNIFIED CME-> 9951-> 9971->Hold | passed | |
| UC862S.CME. U.099 | UNIFIED CME Video Conferencing - Heterogeneous | Verify that Transfer feature is working in Heterogeneous Video Conferencing | 9951-> UNIFIED CME-> 9951-> 9971->Transfer | passed | |
| UC862S.CME. U.101 | UNIFIED CME Video Conferencing - Heterogeneous | Verify that Call Pickup feature is working in Heterogeneous Video Conferencing | 9951-> UNIFIED CME-> 9951-> 9971->Call Pickup | passed | |
| UC862S.CME. U.102 | UNIFIED CME Video Conferencing Homogeneous between 9951,8945 & 9971 | Verify that Video Conferencing is established in Cisco Unified IP Phone 9971,8945 & 9951 | 9951-> Meet-Me DN->UNIFIED CME->9951->8 945 | passed | |
| UC862S.CME. U.103 | UNIFIED CME Video Conferencing Heterogeneous between 9951,8945 & 9971 | Verify that Video Conferencing is established in Cisco Unified IP Phone 9971,8945 & 9951 | 9951-> Meet-Me DN->UNIFIED CME->9951->8 945 | passed | |
| UC862S.CME. U.104 | UNIFIED CME Video Conferencing Homogeneous between 9951,8945 & 9971 -Call Forward | Verify that Video Conferencing is established in Cisco Unified IP Phone 9971,8945 & 9951 | 9951-> UNIFIED CME->9951->9 971->Call Forward | passed | |

Regression

| Project Feature Tested | Total Test Cases | Total % | Passed | Passed % | Passed W/X | Passed W/X % | Failed | Failed % |
|-----------------------------|------------------|---------|--------|----------|------------|--------------|--------|----------|
| Unified CM - Basic | 447 | 20.99 | 447 | 20.99 | 0 | 0 | 0 | 0 |
| Unified CM - Conference | 475 | 22.30 | 475 | 22.30 | 0 | 0 | 0 | 0 |
| Unified CM - Transfer | 230 | 10.80 | 230 | 10.80 | 0 | 0 | 0 | 0 |
| Unified CM - CFWD | 210 | 9.86 | 210 | 9.86 | 0 | 0 | 0 | 0 |
| Unified CM - Park | 206 | 9.67 | 206 | 9.67 | 0 | 0 | 0 | 0 |
| Unified CM - PickUp | 243 | 11.41 | 243 | 11.41 | 0 | 0 | 0 | 0 |
| Unified CM - DND | 106 | 4.98 | 106 | 4.98 | 0 | 0 | 0 | 0 |
| Unified CM - Inter Cluster | 6 | 0.28 | 6 | 0.28 | 0 | 0 | 0 | 0 |
| Unified CM - Hunt List | 3 | 0.14 | 3 | 0.14 | 0 | 0 | 0 | 0 |
| Unified CM - BLF Speed Dial | 5 | 0.23 | 5 | 0.23 | 0 | 0 | 0 | 0 |
| Unified CM - BLF Dpark | 3 | 0.14 | 3 | 0.14 | 0 | 0 | 0 | 0 |
| Unified CM - Barge | 3 | 0.14 | 3 | 0.14 | 0 | 0 | 0 | 0 |
| Unified CM - Meet Me | 193 | 9.06 | 193 | 9.06 | 0 | 0 | 0 | 0 |
| Total | 2130 | 100 | 2130 | 100 | 0 | 0 | 0 | 0 |

| Logical ID | Feature Tested | Case Title | Description | Call Component Flow | Status | Defects |
|-----------------------|----------------|--|---|--|--------|---------|
| UC862F.CUC M.M.010 | CFwd | Unified IP Phone with call forward all to Hunt Pilot number | Unified IP Phone configured with call forward all to a Hunt pilot number. Hunt list configured for the Hunt pilot have two line groups with Top down algorithm. Verify that call routed through all the Unified IP Phones available in the Line Groups. | Unified IP Phone -> Unified CM ->Call forward all -> Hunt pilot number | Passed | |
| UC862F.CUC M.M.011 | CFwdAll | Call to a Unified IP Phone with call forward all to Hunt Pilot number through ICT trunk. | Unified IP Phone configured with call forward all to a Hunt pilot number. Hunt list configured for the Hunt pilot have two line groups with Top down algorithm. Verify that an inter-cluster call from ICT Trunk routed through all the Unified IP Phones available in the Line Groups. | 1.Unified IP Phone A->Unified CM 1->ICT->Unified CM 2->Unified IP Phone B 2.Unified IP Phone B->Forwards-> Unified CM 2->Hunt Pilot. | Passed | |

| Logical ID | Feature Tested | Case Title | Description | Call Component Flow | Status | Defects |
|-----------------------|----------------|---|---|---|--------|---------|
| UC862F.CUC M.M.012 | CFwd | Call to a Unified IP Phone with call forward all to Hunt Pilot number through SIP trunk. | Unified IP Phone configured with call forward all to a Hunt pilot number. Hunt list configured for the Hunt pilot have two line groups with Top down algorithm. Verify that an inter-cluster call from SIP Trunk routed through all the Unified IP Phones available in the Line Groups. | Unified IP Phone1 -> Unified CM -> Call forward All -> Hunt Pilot number -> All Unified IP Phones in line group should ring | Passed | |
| UC862F.CUC M.M.014 | Cfwd | Unified IP Phone with call forward all to Hunt Pilot number with 7925 Unified IP Phone in one of the Line group | Unified IP Phone configured with call forward all to a Hunt pilot number. Hunt list configured for the Hunt pilot have two line groups with Top down algorithm. One of the Line Groups has 7925 Unified IP Phone. Verify that call routed through all the Unified IP Phones and 7925 Unified IP Phone available in the Line Groups. | Unified IP Phone -> Unified CM -> Hunt pilot number | Passed | |

| Logical ID | Feature Tested | Case Title | Description | Call Component Flow | Status | Defects |
|-----------------------|----------------|--|---|---|--------|---------|
| UC862F.CUC M.M.015 | CallPark | Call to a Hunt Pilot number. Attend and park the call - Global Call park | Call to a Hunt Pilot number. Pick the call on any of the Unified IP Phone in the line group. Park the call. Park code assigned regardless of the node. Verify that call gets parked successfully. | 1.Unified IP Phone A->Unified CM->HuntPilot 2.Unified IP Phone B->Answers call->Parks | Passed | |
| UC862F.CUC M.M.016 | CallPark | Call to a Hunt Pilot number. Attend and park the call, Park reversion after reversion timer expires - Global Call park | Call to a Hunt Pilot number. Pick the call on any of the Unified IP Phone in the line group. Park the call. Park code assigned regardless of the node. Verify that park reversion happens successfully after reversion timer expires. | 1.Unified IP Phone A->Unified CM->Hunt Pilot. 2.Unified IP Phone B->Answers Call->Parks the call->Reversion timeout 3.Unified IP Phone B->Answers Call. | Passed | |

| Logical ID | Feature Tested | Case Title | Description | Call Component Flow | Status | Defects |
|-----------------------|--------------------|---|--|---|--------|---------|
| UC862F.CUC M.M.017 | CallPark | Call to a Hunt Pilot number through a ICT trunk. Attend and park the call- Global Call park | Unified IP Phone from Cluster A calls to a Hunt Pilot number in Cluster B through ICT trunk. Pick the call on any of the Unified IP Phone in the line group. Park the call. Park code assigned regardless of the node. Verify that call gets parked successfully. | 1.Unified IP Phone A->Unified CM 1-> ICT ->Unified CM 2->Hunt Pilot. 2.Unified IP Phone B->Answers Call->Parks the call | Passed | |
| UC862F.CUC M.M.018 | Inter Cluster Call | Call to a Hunt Pilot number through a SIP trunk. Attend and park the call-Global Call park | Unified IP Phone from other Cluster A calls to a Hunt Pilot number in Cluster B through SIP trunk. Pick the call on any of the Unified IP Phones in the line group. Park the call. Park code assigned regardless of the node. Verify that call gets parked successfully. | Unified IP Phone1 -> Unified CM 1-> SIP Trunk -> Unified CM2-> Unified IP Phone 2-> Pick the call in any Unified IP Phone registered in groupline -> Park | Passed | |

| Logical ID | Feature Tested | Case Title | Description | Call Component Flow | Status | Defects |
|-----------------------|----------------|---|---|---|--------|---------|
| UC862F.CUC M.M.019 | CallPark | Call to a Hunt Pilot with Broadcast Algorithm selected in the Line Group. Attend and park the call - Global Call park | Call to a Hunt Pilot number. Pick the call on 69XX Unified IP Phones in the line group. Park the call. Park code assigned regardless of the node. Verify that call gets parked successfully. | 1.Unified IP Phone A->Unified CM ->Hunt Pilot. 2.Unified IP Phone B(69XX)->Answers Call->Parks the call. | Passed | |
| UC862F.CUC M.M.020 | CallPark | Call to a Hunt Pilot with Broadcast Algorithm selected in the Line Group. Attend and park the call. - Global Call park | Call to a Hunt Pilot number. Pick the call on 99XX Unified IP Phones in the line group. Park the call. Park code assigned regardless of the node. Verify that call gets parked successfully. | Unified IP Phone -> Unified CM -> Hunt pilot number -> Attend in Unified IP Phone-> Park | Passed | |
| UC862F.CUC M.M.021 | CallPark | Call to a Hunt Pilot with Broadcast Algorithm selected in the Line Group. Park the call and Park reversion after reversion timer expires - Global Call park | Call to a Hunt Pilot number. Pick the call on 69XX Unified IP Phone in the line group. Park the call. Park code assigned regardless of the node. Verify that park reversion happens successfully after reversion timer expires. | 1.Unified IP Phone A->Unified CM->HuntPilot 2.Unified IP Phone B(69XX)->Answers call->Parks->R eversion timeout->Unifie d IP Phone B rings. | Passed | |

| Logical ID | Feature Tested | Case Title | Description | Call Component Flow | Status | Defects |
|-----------------------|--------------------|---|---|--|--------|---------|
| UC862F.CUC M.M.022 | CallPark | Call to a Hunt Pilot with Broadcast Algorithm selected in the Line Group. Park the call and Park reversion after reversion timer expires - Global Call park | Call to a Hunt Pilot number. Pick the call on 99XX Unified IP Phone in the line group. Park the call. Park code assigned regardless of the node. Verify that park reversion happens successfully after reversion timer expires. | Unified IP Phone -> Unified CM -> Hunt pilot number -> Attend in Unified IP Phone->Park-> Check Park reversion | Passed | |
| UCJ862FN.CU CM.001 | Inter Cluster Call | Redirect call across 3 Intercluster trunk disconnected when multiple hold/resume. | Verify the ability to redirect call across 3 Intercluster trunks when it is disconnected while the call is still on multiple hold/resume. | 1. Unified IP Phone A ->Unified CM 1-> ICT -> Unified CM 2 -> Unified IP Phone B 2. Unified IP Phone A -> Hold ->Unified IP Phone B 3. Unified IP Phone A -> Resume ->Unified IP Phone B 4. Unified IP Phone B -> Unified CM 2 -> ICT ->Unified CM 3 -> Unified IP Phone B 5. Unified IP Phone B -> Hold ->Unified IP Phone C 6. Unified IP Phone B -> Resume ->Unified IP Phone C | Passed | |

| Logical ID | Feature Tested | Case Title | Description | Call Component Flow | Status | Defects |
|-----------------------|----------------|---|--|---|--------|---------|
| UCJ862FN.CU CM.002 | Basic | IP-to-IP Intra-cluster Single-site Shared-line Appearance | Verify that shared-line appearance works in a single-site scenario | 1. Unified IP Phone A ->Unified CM -> Unified IP Phone B (Line 2)->No Answer 2. Unified IP Phone 3 (Line 2)-> pickUp ->Unified IP Phone 2 | Passed | |
| UCJ862FN.CU CM.003 | Basic | Checking Ring Setting of Busy Station Policy in Shared line | Verify that ring setting of Busy Station Policy in Shared line with any exceptions | 1. SCCP Phone4 -> Unified CM -> SCCP Phone1 ->Answer call 2. SCCP Phone 5-> Unified CM ->SCCP Phone1-> Answercall (Holds the first call) | Passed | |
| UCJ862FN.CU CM.004 | Auto Pickup | Auto Pickup feature in a shared line | Verify AutoPickup in shared Line | 1. Unified IP Phone A -> Unified CM -> Unified IP Phone B ->No Answer 2. Unified IP Phone C-> Unified CM -> autoPickUp -> Unified IP Phone B. | Passed | |

| Logical ID | Feature Tested | Case Title | Description | Call Component Flow | Status | Defects |
|-----------------------|----------------|--|---|---|--------|---------|
| UCJ862FN.CU CM.005 | BLF Dpark | Assisted BLF Dpark on a transferred call | Verify when call is parked using Assisted BLF Dpark and the parkee will transfer the call to 3rd party the RT phone try to retrieve the call | 1. Unified IP Phone A -> Unified CM -> Unified IP Phone B -> Answer Call. 2. Unified IP Phone A-> Unified CM -> Call Park. 3. Unified IP Phone B -> Unified CM -> Transfer -> Unified IP Phone C. 4. Unified IP Phone A -> Unified CM -> Retrive Park -> Unified IP Phone C. | Passed | |
| UCJ862FN.CU CM.006 | Transfer | BLF (Busy Line Field): Verify Busy Line Field speed dial combining Call blind transfer | EpFeature (feature applied endpoint) speed dial epAsstnt1. epAsstnt1 (assistant endpoint 1) blind transfer the call to epAsstnt2 (assistant endpoint 2) by transfer softkey. EpFeature and epAsstnt2 talk directly and epAsstnt1 dropped from the call. | 1. Unified IP Phone A -> Unified CM -> BLF -> Unified IP Phone B 2. Unified IP Phone A -> Unified CM -> Transfer -> Unified IP Phone C (Using Softkey). | Passed | |

| Logical ID | Feature Tested | Case Title | Description | Call Component Flow | Status | Defects |
|-----------------------|----------------|--|--|---|--------|---------|
| UCJ862FN.CU CM.007 | BLF speed dial | BLF (Busy Line Field): Verify Busy Line Field speed dial combining hold and resume feature | EpFeature (feature applied endpoint) pushes the BLF button to speed dial epAsstnt1 (assistant endpoint 1). During the call Hold/resume either 3 times. | Unified IP Phone1 ->Speed Dial -> Unified IP Phone2 ->Hold and resume (Three times) | Passed | |
| UCJ862FN.CU CM.008 | Auto Pickup | SCCP Shared Line Phone Auto Call Pickup | Verify auto call pickup on a SCCP phone with shared lines. | Unified IP Phone1 -> CM -> Unified IP Phone2 -> autoPickup ->Unified IP Phone3 (Line 2) | Passed | |
| UCJ862FN.CU CM.010 | Basic | Roles displaying in Groups area. | Verify that roles displaying in Groups area. The screen should be navigated back to Roles Assignment page. | | Passed | |
| UCJ862FN.CU CM.011 | Basic | Unified CM Showing Blank page with user ID while refreshing the application | Verify Unified CM Showing Blank page with user ID while refreshing the application | | Passed | |
| UCJ862FN.CU CM.012 | Hunt List | Hunt List config page hangs while modifying Hunt List name. | Verify that Hunt List config page hangs while modifying Hunt List name. | | Passed | |
| UCJ862FN.CU CM.013 | Basic | Unified CM showing the wrong status for the registered Unified IP Phones | Verify Unified CM showing the wrong status for the registered Unified IP Phones | | Passed | |

| Logical ID | Feature Tested | Case Title | Description | Call Component Flow | Status | Defects |
|-----------------------|----------------|---|---|---|--------|---------|
| UCJ862FN.CU CM.014 | Barge | Secured Single button Barge across SIP Trunk | The objective of the test to verify secured Single Button Barge across SIP Trunk with endpoints encrypted | Unified IP Phone A -> Unified CM -> SIP Trunk ->Unified IP Phone B -> Unified IP phone C (press shared line to barge) -> Phone A disconnect | Passed | |
| UCJ862FN.CU CM.015 | CFwdAll | Unified IP Phone with call forward all to Hunt Pilot number with 7925 Unified IP Phone in one line group. | Unified IP Phone configured with call forward all to a Hunt pilot number. Hunt list configured for the Hunt pilot have two line groups with Top down algorithm. One of the Line Groups has 7925 Unified IP Phone. Verify that call routed through all the Unified IP Phones and 7925 Unified IP Phone available in the Line Groups. | Unified IP Phone H -> Unified IP Phone G, Unified IP Phone G-> Call Fwd All -> Hunt Pilot Number | Passed | |
| UCJ862FN.CU CM.016 | CallPark | Call revert to configured DN, after the reversion timer expires | Verify that a call reverts to configured DN successfully after reversion timer expires | Unified IP Phone A -> Unified CM -> Unified IP Phone B -> park -> 30 seconds -> Unified IP Phone B | Passed | |

| Logical ID | Feature Tested | Case Title | Description | Call Component Flow | Status | Defects |
|-----------------------|----------------|---|---|---|--------|---------|
| UCJ862FN.CU CM.017 | Transfer | Ring back tone with Unified IP Phones during call transfer scenario | Ensure that two Unified IP Phones, one SCCP phone, Two Clusters Call is made from Unified IP Phone A to Unified IP Phone B; Phone B transfers call to Phone C. Phone A should hear ring back tone. | Unified IP Phone A -> Unified CM -> Unified IP Phone B -> Transfer -> Unified IP Phone C | Passed | |
| UCJ862FN.CU CM.018 | BLF speed dial | Verify Busy Line Field speed dial combining ad-hoc conference by using Join Softkey | epFeature (feature applied endpoint) pushes the BLF button to speed dial epAsstnt1 (assistant endpoint 1). epFeature pushes another BLF button to speed dial epAsstnt2 (assistant endpoint 2) as second call. EpFeature join softkey to have a three party conference call. | 1. Unified IP Phone A -> Unified CM -> BLF (Busy Line Field) -> Unified IP Phone B 2. Unified IP Phone A -> Unified CM -> BLF (Busy Line Field) -> Unified IP Phone C 3. Unified IP Phone A -> Unified CM -> Conference (Using Join Softkey). | Passed | |

| Logical ID | Feature Tested | Case Title | Description | Call Component Flow | Status | Defects |
|-----------------------|----------------|--|---|---|--------|---------|
| UCJ862FN.CU CM.020 | BLF speed dial | BLF (Busy Line Field): Verify Busy Line Field speed dial combining call direct transfer. | epFeature (feature applied endpoint) pushes the BLF button to speed dial epAsstnt1 (assistant endpoint 1). epFeature pushes another BLF button to speed dial epAsstnt2 (assistant endpoint 2) as second call. EpFeature direct transfer these 2 calls | 1. Unified IP Phone A -> Unified CM -> BLF (Busy Line Field) -> Unified IP Phone B 2. Unified IP Phone A -> Unified CM -> BLF (Busy Line Field) -> Unified IP Phone C 3. Unified IP Phone A -> Unified CM -> Conference (Using Join Softkey). | Passed | |
| UCJ862FN.CU CM.021 | Basic | Modified presence server settings are not getting updated | The status message says that the configuration updated successfully but when we go back and verify above steps, we could not see the modified settings. | 1. Unified IP Phone A -> Unified CM -> BLF (Busy Line Field) -> Unified IP Phone B 2. Unified IP Phone A -> Unified CM -> BLF (Busy Line Field) -> Unified IP Phone C 3. Unified IP Phone A -> Unified CM -> Direct Transfer. | Passed | |

| Logical ID | Feature Tested | Case Title | Description | Call Component Flow | Status | Defects |
|-----------------------|-----------------------|---|--|---|--------|---------|
| UCJ862FN.CU CM.022 | Inter Cluster Call | IP-to-IP Inter-cluster Call for Busy Line | Verify that basic inter-cluster call works for busy line | Unified IP Phone A--> Unified CM 1--> Unified IP Phone B Unified IP Phone C->Unified CM 2->ICT-->Unifi ed CM 1 ->unified Unified IP PhoneB | Passed | |
| UCJ862FN.CU CM.023 | CallPark | Park reversion done on a Shared line by a non-DND enabled phone | Ensure that park reversion done on a Shared line by a non-DND enabled phone | Unified IP Phone A-> Unified CM 1-> Unified IP Phone B (line 2) Unified IP Phone B->parks->Afte r reversion timeout-> Unified IP Phone B answers call | Passed | |
| UCJ862FN.CU CM.024 | Conference | SCCP Phone Shared Line Conference Call and Call Transfer | Verify conference calls and call transfer on a SCCP phone with shared lines | | Passed | |
| UCJ862FN.CU CM.025 | Auto Pickup | Checking auto pickup feature in Shared line | Checking auto pickup feature in Shared line in SCCP Unified IP Phone | Unified IP Phone D-> Unified CM ->Sharedline number Unified IP Phone A->AutoPickup | Passed | |
| UCJ862FN.CU CM.026 | Basic | Switchmode never option not available in H.323 fallback | Verify that Switchmode never option not available in H.323 fallback | | Passed | |

| Logical ID | Feature Tested | Case Title | Description | Call Component Flow | Status | Defects |
|-----------------------|----------------------|---|--|---|--------|---------|
| UCJ862FN.CU CM.027 | Basic | Phone BTN Template 9971 list all PHN Templates | Verify that Phone BTN Template 9971 list all PHN Templates | | Passed | |
| UCJ862FR.CU CM.001 | CFwd Unregistered | Display status on 69XX Unified IP Phone when call forwarded through CFUR | Verify that destination Unified IP Phone displays like Forwarded for XXXX (XXXX is the DN of Unregistered phone) when call comes through CFUR | Unified IP Phone A -> Unified CM -> Unified IP Phone C -> Call forward -> Unified IP Phone B | Passed | |
| UCJ862FR.CU CM.002 | CFwdAll | Forward call in 69XX Unified IP Phone using Call Forward All feature through trunk | Verify that call is forwarding successfully in 69XX Unified IP Phone using Call Forward All feature from one Unified CM to another Unified CM through Trunk and display status on destination Unified IP Phone shows correctly | Unified IP Phone A ->Unified CM -> Unified IP Phone B -> Call Forward -> SIP trunk -> Unified CM2 -> Unified IP Phone C | Passed | |
| UCJ862FR.CU CM.003 | CFwdAll | Forward call using Call Forward All feature through ICT trunk | Verify that call is forwarding successfully using Call Forward All feature from one Unified CM to another Unified CM through ICT Trunk | Unified IP Phone A ->Unified CM -> Unified IP Phone B -> Call Forward -> ICT trunk -> Unified CM2 -> Unified IP Phone C | Passed | |

| Logical ID | Feature Tested | Case Title | Description | Call Component Flow | Status | Defects |
|-------------------------|-----------------------|--|---|---|--------|---------|
| UCJ86IF_SR.C UCM.001 | Inter Cluster Call | Intercluster audio call between 9951/71 and 7921 wireless Unified IP Phone | Verify that users able to listen audio clearly when call is established between 9951/71 and 7921 Unified IP Phones | Unified IP Phone A(9951)->Unifi ed CM ->Unified IP Phone B(7921) | Passed | |
| UCJ86IF_SR.C UCM.002 | Inter Cluster Call | Intercluster audio call between 9951/71 and 7925 wireless Unified IP Phone | Verify that users able to listen audio clearly when call is established between 9951/71 and 7925 Unified IP Phones | Unified IP Phone1 -> Unified CM -> Unified IP Phone 2 | Passed | |
| UCJ86IF_SR.C UCM.006 | Conference | Intercluster Conference call between 9951, 7921 and 7925 Unified IP Phones | Verify that the intercluster conference call established successfully between 9951/71, 7921 and 7925 Unified IP Phones | 1.Unified IP Phone A->Unified CM 1->Unified IP Phone B 2.Unified IP Phone A->ConfBegin- >Unified CM 1->SIP trunk ->Unified CM 2->Unified IP Phone C | Passed | |
| UCJ86IF_SR.C UCM.008 | Conference | Intercluster Conference call between 9951, 9971 and 7925 Unified IP Phones | Verify that the intercluster conference call established successfully between 9951, 9971 and 7925 Unified IP Phones | Unified IP Phone A -> Unified CM -> Unified CM -> Unified IP Phone B -> call connected -> Conference -> Unified IP Phone C | Passed | |

| Logical ID | Feature Tested | Case Title | Description | Call Component Flow | Status | Defects |
|-------------------------|----------------|--|---|---|--------|---------|
| UCJ86IF_SR.C UCM.009 | CFwdAll | Call Forward All to 7925 Unified IP Phone | Verify that all calls coming to 7921 Unified IP Phone successfully forwarding to DN of Unified IP Phone which is configured in 7921 Unified IP Phone Call Forward All field | Unified IP Phone A->Unified CM->Unified IP Phone B->CFwdAll-> Unified CM->Unified IP Phone C | Passed | |
| UCJ86IF_SR.C UCM.010 | CFwdAll | Call Forward All to Analog Phone | Verify that all calls coming to 7921 Unified IP Phone successfully forwarding to DN of Unified IP Phone which is configured in 7921 Unified IP Phone Call Forward All field | 1.Unified IP Phone A->Unified CM ->Unified IP Phone B(7921). 2.Unified IP Phone B->CFwdAll-> Unified CM->Unified IP Phone C | Passed | |
| UCJ86IF_SR.C UCM.011 | Basic | SIP TRUNK config between Unified CM and Unified CM2 | Verify that whether SIP trunk configured successfully without any errors related to memory allocation and users able to communicate with each other | Unified IP Phone A -> Unified CM -> SIP trunk -> Unified CM2 -> Unified IP Phone B | Passed | |
| UCJ86IF_SR.C UCM.012 | Basic | Adding/updatin g Phone DN page | Verify that user able to add/update Phone DN page without any warning or error messages | | Passed | |

| Logical ID | Feature Tested | Case Title | Description | Call Component Flow | Status | Defects |
|-------------------------|----------------|---|---|--|--------|---------|
| UCJ86IF_SR.C UCM.013 | Do Not Disturb | DND feature (Call Reject) for Unified IP Phone 9951 | Verify that whether DND option call reject works fine when user calls Unified IP Phone 9951 | Unified IP Phone A -> Unified CM -> Unified IP Phone B(9951) | Passed | |
| UCJ86IF_SR.C UCM.014 | Do Not Disturb | DND option (Ringer Off) for Unified IP Phone 9951 | Verify that whether DND option Ringer Off works fine when user calls Unified IP Phone 9951 | Unified IP Phone A->Unified CM ->Unified IP Phone B(9951). | Passed | |
| UCJ86IF_SR.C UCM.021 | CFwdAll | Call Forward All feature on 9951 Unified IP Phone | Verify that calls coming from Unified IP Phone to 9951 Unified IP Phone forwarding to DN of 6921 Unified IP Phone which was assigned in Call Forward All column | 1.Unified IP Phone A -> Unified CM -> Unified IP Phone B(9951) 2.Unified IP Phone B -> CFwdAll -> Unified CM -> Unified IP Phone C(6921). | Passed | |
| UCJ86IF_SR.C UCM.022 | CFwd | Call Forward All feature on 6961 Unified IP Phone | Verify that calls coming from Unified IP Phone to 6961 Unified IP Phone forwarding to DN of 7921 Unified IP Phone which was assigned in Call Forward All column | Unified IP Phone1 -> Unified CM -> Unified IP Phone2 -> Forward -> DN assigned in Call Forward All | Passed | |

| Logical ID | Feature Tested | Case Title | Description | Call Component Flow | Status | Defects |
|-------------------------|----------------|--|---|--|--------|---------|
| UCJ86IF_SR.C UCM.024 | Basic | Ringtone and Audio on 7921 Unified IP Phone | Verify that when call comes from any Unified IP Phone to 7921 Unified IP Phone ringtone coming perfectly and after attend the call user able to listen and talk | Unified IP Phone A -> Unified CM -> Unified IP Phone B | Passed | |
| UCJ86IF_SR.C UCM.025 | Basic | On-hook dialing from 69XX Unified IP Phone when maximum number of calls set to 2 | Verify that on-hook dialing happens from 69XX Unified IP Phone without any errors when maximum number of calls set to 2 | 1.Unified IP Phone A -> Presses speaker button 2.Unified IP Phone A -> Unified CM -> Unified IP Phone B(69XX) | Passed | |
| UCJ86IF_SR.C UCM.027 | Basic | Hold and calling another number on 69XX Unified IP Phone when maximum number of calls set to 2 | Verify the status for 69XX Unified IP Phone in the situation hold and calling another number when maximum number of calls set to 2 | 1.Unified IP Phone A -> Unified CM -> Unified IP Phone B 2.Unified IP Phone B -> Answers call -> Hold the call 3.Unified IP Phone B -> Unified CM ->Unified IP Phone C | Passed | |
| UCJ86IF_SR.C UCM.028 | Basic | Hold and calling another number on 99XX Unified IP Phone when maximum number of calls set to 2 | Verify the status for 99XX Unified IP Phone in the situation hold and calling another number when maximum number of calls set to 2 | Unified IP Phone 1 -> Unified CM -> Unified IP Phone 2 -> Hold -> Unified IP Phone 2 call 2 -> Unified IP Phone 3 | Passed | |

| Logical ID | Feature Tested | Case Title | Description | Call Component Flow | Status | Defects |
|-------------------------|----------------------|---|--|---|--------|---------|
| UCJ86IF_SR.C UCM.029 | CFwd | CFUR (Call forward Unregistered) enable on 69XX Unified IP Phone as DN of any Unified IP Phone | Verify that call comes to 69XX Unified IP Phone which is in unregistered state forwarded to DN of Unified IP Phone mentioned in the CFUR column | 1.Unified IP Phone A -> Unified CM -> Unified IP Phone B(69XX) 2.Unified IP Phone B -> Forwards call -> Unified CM->IP Phone C | Passed | |
| UCJ86IF_SR.C UCM.030 | CFwd Unregistered | Display status on Unified IP Phone when call forwarded through CFUR | Verify that destination Unified IP Phone forward a call comes through CFUR | Unified IP Phone A -> Unified CM -> Unified IP Phone B -> CFUR -> Unified CM -> Unified IP Phone C | Passed | |
| UCJ86IF_SR.C UCM.031 | CFwd | 69XX Unified IP Phone with CFUR(Call forward Unregistered) feature | Verify that when call comes to unregistered 69XX Unified IP Phone which enables CFUR forwards to all Unified IP Phones listed in hunt pilot list | Unified IP Phone -> Unified CM -> Unregistered Unified IP Phone -> CFUR -> Unified IP Phones in Hunt pilot number | Passed | |
| UCJ86IF_SR.C UCM.032 | CFwd | CFUR (Call forward Unregistered) enable on 69XX Unified IP Phone as Hunt pilot number - (No Config) | Verify that call comes to 69XX Unified IP Phone which is in unregistered state forwarded to Hunt pilot number | 1.Unified IP Phone A -> Unified CM -> Unified IP Phone B(69XX) 2.Unified IP Phone B -> Forwards -> Unified CM -> Hunt Pilot. | Passed | |

| Logical ID | Feature Tested | Case Title | Description | Call Component Flow | Status | Defects |
|-------------------------|----------------------|--|---|---|--------|---------|
| UCJ86IF_SR.C UCM.033 | CFwd Unregistered | Forwarding twice the call to destination Unified IP Phone when CFUR (Call forward Unregistered) feature is in progress | Verify that call comes to Phone B which has enabled CFUR forward to Phone C which also has enable CFUR forward to DN of Unified IP Phone mentioned in Phone C CFUR column | 1.Unified IP Phone A -> Unified CM -> Unified IP Phone B 2.Unified IP Phone B -> CFUR -> Unified CM -> Unified IP Phone C -> CFUR -> Unified CM -> Unified IP Phone D. | Passed | |
| UCJ86IF_SR.C UCM.035 | CFwd | Forward Busy feature on 99XX Unified IP Phone when Busy Trigger set as 2 | Verify that when call comes to 99XX Unified IP Phone which is in busy state will forward call to DN of Unified IP Phone mentioned in the forward busy column of 99XX Unified IP Phone | Unified IP Phone A -> Unified CM -> Unified IP Phone B(99xx) -> Forwards -> Unified CM -> Unified IP Phone C. | Passed | |
| UCJ86IF_SR.C UCM.044 | Transfer | Consultative transfer to Cisco IP Communicator | Verify that consultative transfer happens successfully between 6961 Unified IP Phone, 9971 Unified IP Phone and Cisco IP Communicator | Unified IP Phone1 -> Unified CM -> Unified IP Phone 2 -> call connected -> Consultative transfer -> IP Communicator | Passed | |

■ Regression

| Logical ID | Feature Tested | Case Title | Description | Call Component Flow | Status | Defects |
|--------------------------|---------------------|--|---|---|--------|---------|
| UCJ862F.AUT O.II.0001 | CallForward All | CallForward All feature for Cisco IP Communicator | Verify that all Calls coming from Cisco Unified IP Phone A to Cisco IP Communicator is successfully forwarded to Cisco Unified IP Phone B | Cisco Unified IP Phone A -> Unified CM -> CIPC B -> CallForward All -> Cisco Unified IP Phone C | Passed | |
| UCJ862F.AUT O.II.0172 | Adhoc Conference | Ad hoc conference by using Conf Softkey. | EpAsstnt1 (assistant endpoint 1) makes 1st call to epFeature (feature applied endpoint). EpAsstnt2 (assistant endpoint 2) makes 2nd call to epFeature. epFeature join these 2 calls as a conference call. | Cisco Unified IP Phone A -> Unified CM -> Cisco Unified IP Phone B ; Cisco Unified IP Phone C -> Unified CM -> Cisco Unified IP Phone B; Cisco Unified IP Phone B-> Join Call | Passed | |

| Logical ID | Feature Tested | Case Title | Description | Call Component Flow | Status | Defects |
|--------------------------|--|--|--|--|--------|---------|
| UCJ862F.AUT O.II.0268 | Adhoc Conference | Ad-hoc Inter-cluster Conference with 6 Participants | Verify that in an IP-to-PSTN ad-hoc intra-cluster conference with 6 participants, last 2 can drop without impacting the others. | Cisco Unified IP Phone A -> Unified CM -> Cisco Unified IP Phone B ; Cisco Unified IP Phone B -> Conf -> Cisco Unified IP Phone C; Cisco Unified IP Phone B-> Conf -> Cisco Unified IP Phone D; Cisco Unified IP Phone B-> Conf -> Cisco Unified IP Phone E; Cisco Unified IP Phone B-> Conf -> Cisco Unified IP Phone F | Passed | |
| UCJ862F.AUT O.II.0275 | Call between two Cisco Unified IP Phones (SCCP) | Call Between a Cisco Unified IP Phone and an SCCP Phone | Verify that a Cisco Unified IP Phone registered to a Cisco Unified communication s Manager can call to an Unified IP Phone 796X (SCCP) | Cisco Unified IP Phone A -> Unified CM -> Cisco Unified IP Phone B | Passed | |
| UCJ862F.AUT O.II.0307 | Call Pickup | Directed Call Pickup when multiple calls are available for pickup, Check the Call Pick up Group Notification Settings. | Ensure directed Call Pickup when multiple calls are available for pickup and Check the Call Pick up Group Notification Settings. | Cisco Unified IP Phone A -> Unified CM -> Cisco Unified IP Phone B, Cisco Unified IP Phone C -> Gpickup | Passed | |

| Logical ID | Feature Tested | Case Title | Description | Call Component Flow | Status | Defects |
|--------------------------|---------------------|---------------------------------------|--|--|--------|---------|
| UCJ862F.AUT O.II.0532 | Adhoc Conference | Ad-hoc Intra-cluster Conference | Verify that all parties in an ad-hoc intra-cluster conference remain connected when the non-conference initiator drops the call | Cisco Unified IP Phone A -> Unified CM -> Cisco Unified IP Phone B; Cisco Unified IP Phone A -> Conf -> Cisco Unified IP Phone C | Passed | |
| UCJ862F.AUT O.II.0541 | DND | DND Feature of CIPC | Verify by making a call from a Cisco Unified IP Phone to an Unified IP Phone (SCCP) that transfers the call to an IP Communicator with Do Not Disturb (DND) feature activated. | Cisco Unified IP Phone A -> Unified CM -> Cisco Unified IP Phone B -> transfer -> CIPC C | Passed | |

Related Documentation

Cisco VCS Control

Cisco TelePresence Basic Cisco VCS configuration Deployment Guide

http://www.cisco.com/en/US/docs/telepresence/infrastructure/vcs/config_guide/Cisco_VCS_Basic_Configuration_Cisco_VCS_Control_with_Cisco_VCS_Expressway_Deployment_Guide_X7-0.pdf

Cisco Unified Communications Manager v6.1, 7 and 8 - Cisco TelePresence Deployment Guide

http://www.cisco.com/en/US/docs/telepresence/infrastructure/vcs/config_guide/Cisco_VCS_Cisco_Unified_Communications_Manager_Deployment_Guide_CUCM_6-1_7_8_and_X5-1.pdf

Cisco Virtualization Experience Infrastructure

Cisco Virtualization Experience Infrastructure CVD for VMware View

http://www.cisco.com/en/US/docs/solutions/Enterprise/Data_Center/VXI/CVD/VXI_CVD_VMware.html

Cisco Virtualization Experience Infrastructure (VXI) Configuration Guide

http://www.cisco.com/en/US/docs/solutions/Enterprise/Data_Center/VXI/configuration/VXI_Config_Guide.pdf

Cisco Cius**Cisco Cius Administration Guide**

http://www.cisco.com/en/US/docs/voice_ip_comm/ccmcd/admin/9_2_1/english/ciusag921.pdf

Cisco Unified Communications Manager**Documentation Guide**

http://www.cisco.com/en/US/docs/voice_ip_comm/cucm/docguide/8_6_2/dg862a.html

Software Compatibility Matrix

http://www.cisco.com/en/US/docs/voice_ip_comm/cucm/compat/ccmcompmatr.html

IPv6 Deployment Guide

http://www.cisco.com/en/US/docs/voice_ip_comm/cucm/srnd/ipv6/ipv6srnd.html

Cisco Unified Communications Manager on Virtualized Servers

http://www.cisco.com/en/US/docs/voice_ip_comm/cucm/virtual/servers.html

Cisco Unified Communications Manager Documentation Guide for Release 8.6(2)

http://www.cisco.com/en/US/docs/voice_ip_comm/cucm/docguide/8_6_2/dg862.html

New and Changed Information of Cisco Unified Communications Manager Release 8.6(2)

http://www.cisco.com/en/US/docs/voice_ip_comm/cucm/rel_notes/8_6_2/delta/delta862.html

Installation Guide for Cisco Unified Presence Release 8.6

http://www.cisco.com/en/US/docs/voice_ip_comm/cups/8_6/english/install_upgrade/installation/guide/CUP_8_6_InstallGuide_Final_2.pdf

Cisco Unified Communications Manager Express System Administrator Guide

http://www.cisco.com/en/US/docs/voice_ip_comm/cucme/admin/configuration/guide/cmeinter.html

Cisco Jabber for iPhone Administration Guide

http://www.cisco.com/en/US/docs/voice_ip_comm/jabber/iPhone/8.6/JABI_BK_J29330BB_00_jabber-for-iphone-admin-guide_chapter_011.html#JABI_TK_SE559903_00

Cisco Jabber for Mac Installation and Configuration Guide

http://www.cisco.com/en/US/docs/voice_ip_comm/jabber/mac/8.6/b_jabber_mac_admin_guide.pdf

