



Oracle Enterprise Session Border Controller –
Acme Packet 4600 and Cisco Unified
Communications Manager 10.5 for Enterprise
SIP Trunking with NTT Communications

Technical Application Note



Disclaimer

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Intended Audience

This document is intended for use by Oracle Systems Engineers, third party Systems Integrators, Oracle Enterprise customers and partners and end users of the Oracle Enterprise Session Border Controller (E-SBC). It assumes that the reader is familiar with basic operations of the Oracle Enterprise Session Border Controller 4600 platform.

Document Overview

This Oracle technical application note outlines the recommended configurations for the Oracle enterprise session border controller AP-4600 for connecting NTT Communications SIP Trunking service to Cisco Unified Call Manager (CUCM) 10.5 customers. The solution contained within this document has been certified using Oracle's Acme Packet OS ECZ 7.5 software

Cisco Unified Call Manager provides industry-leading reliability, security, scalability, efficiency, and enterprise call and session management and is the core call control application of the collaboration portfolio. This reduces the cost and complexity of extending an enterprise's telephony system outside its network borders. Oracle Enterprise Session Border Controllers (SBCs) play an important role in SIP trunking as they are used by many ITSPs and some enterprises as part of their SIP trunking infrastructure.

This application note has been prepared as a means of ensuring that SIP trunking between CUCM 10.5, Oracle E-SBCs and IP Trunking services are configured in the optimal manner.

It should be noted that while this application note focuses on the optimal configurations for the Oracle ESBC in an enterprise CUCM 10.5 environment, the same SBC configuration model can also be used for other enterprise SIP trunking applications with a few tweaks to the configuration for required features. In addition, it should be noted that the SBC configuration provided in this guide focuses strictly on the CUCM Server associated parameters. Many SBC applications may have additional configuration requirements that are specific to individual customer requirements. These configuration items are not covered in this guide. Please contact your Oracle representative with any questions pertaining to this topic.

For additional information on CUCM 10.5, please visit <http://www.cisco.com/c/en/us/products/unified-communications/unified-communications-manager-version-10-5/index.html>

Introduction

Audience

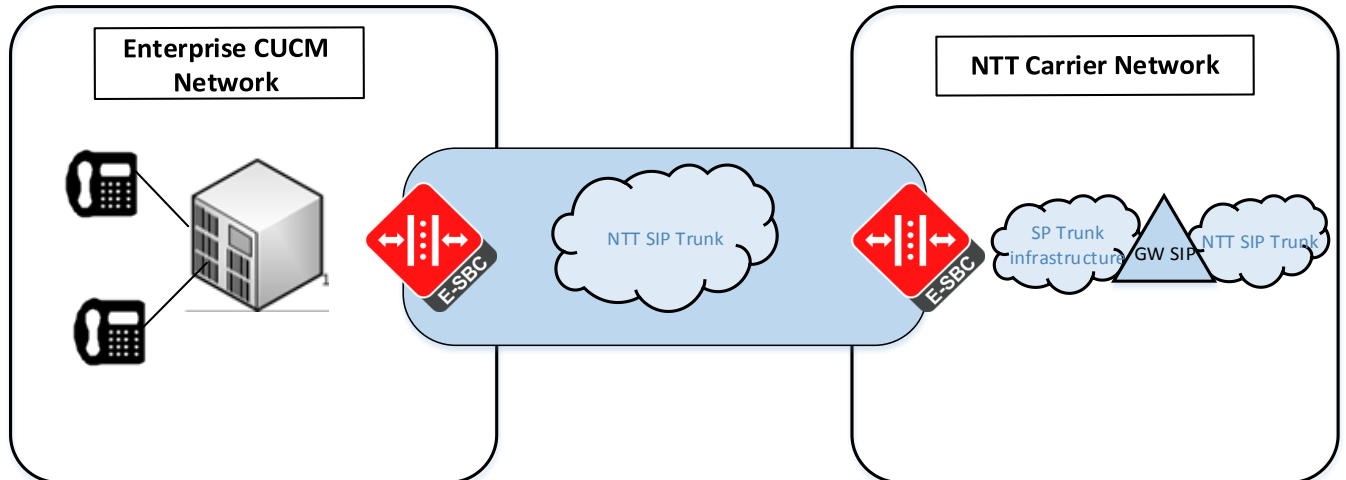
This is a technical document intended for telecommunications engineers with the purpose of configuring the Oracle Enterprise SBC and CUCM 10.5. There will be steps that require navigating the CUCM 10.5 server configuration as well as the Acme Packet Command Line Interface (ACLI). Understanding the basic concepts of TCP/UDP, IP/Routing, and SIP/RTP are also necessary to complete the configuration and for troubleshooting, if necessary.

Requirements

- Fully functioning Cisco UCM 10.5
- Cisco hard phones connected/registered to the CUCM server
- Oracle Enterprise Session Border Controller (hereafter Oracle E-SBC) 4600 series running ECZ750.
- Oracle E-SBC having established SIP connectivity with CUCM on CPE side and NTT SIP trunk on PSTN side.

Architecture

The following reference architecture shows a logical view of the connectivity between CUCM and the SBC.



Area on the left is the customer's on premise infrastructure, which includes the CUCM with the enterprise phones systems. Area on the right represents the NTT communications infrastructure which provides PSTN service via the SIP trunk. The SBC provides integration of these two environments over an IP network and provides security, service reachability, interoperability/normalization of SIP messages over the IP network. The CUCM and E-SBC are the edge components that form the boundary of the SIP trunk.

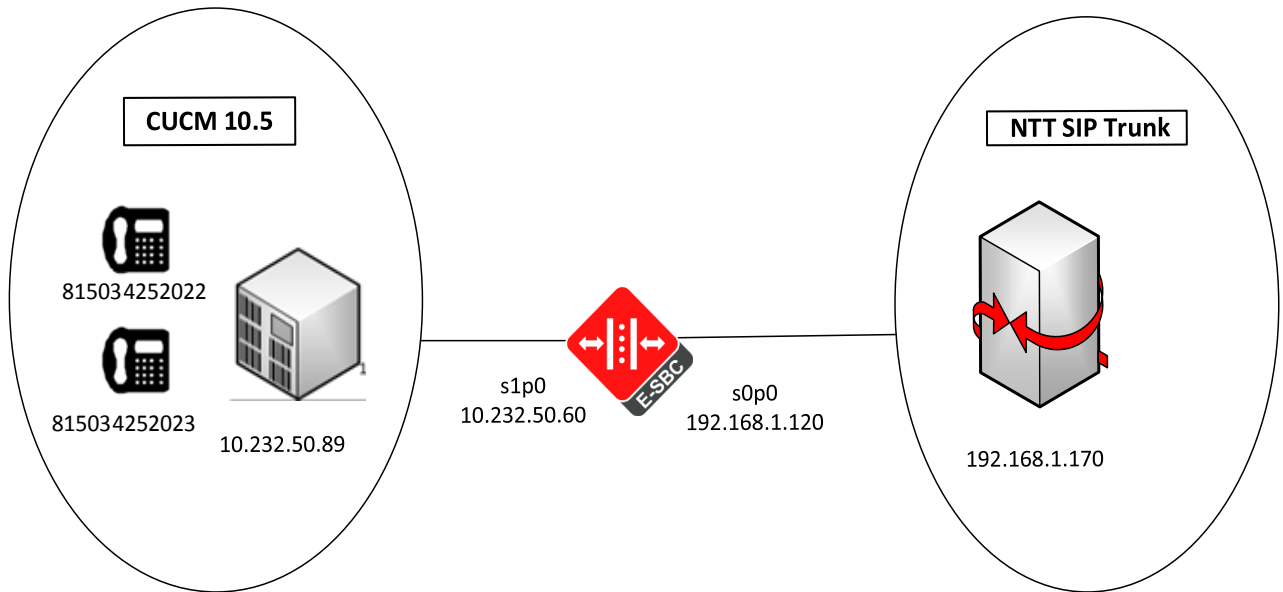
As per the NTT network requirements, the customer devices are required to register and support authentication. The Oracle ESBC supports authentication and performs registration on behalf of the CUCM.

The configuration, validation and troubleshooting of these two is the focus of this document and will be described in two phases:

- Phase 1 – Configuring the Cisco Unified Call Manager v10.5
- Phase 2 – Configuring the Oracle E-SBC

Lab Configuration

The following diagram, similar to the Reference Architecture described earlier in this document, illustrates the lab environment created to facilitate certification testing.



Phase 1 – Configuring the Cisco Unified Call Manager v10.5

The enterprise will have a fully functioning CUCM v10.5 installed and deployed for this certification.

There are a few parts for configuring CUCM v10.5 to be configured and connected to operate with the Oracle E-SBC:

- Creating a SIP profile in CUCM and enabling OPTIONS ping to pro-actively monitor the SIP connectivity with the SBC
- Adding the SBC as a trunk to the CUCM infrastructure
- Creating a route pattern in the CUCM configuration to utilize the configured SBC trunk and route calls from CUCM to the SBC
- Additional configuration to add Directory Numbers, Phones to register to the CUCM and enabling a DHCP server for assigning IP addresses to Cisco phones
- NTT communications requires G.729 for certain tests on the trunk side, configuration to support G.729 for the phones in CUCM

To add the SBC as a trunk in CUCM, we will need:

- IP address of the CUCM NIC facing SBC
- IP address and port of the sip interface of the SBC facing CUCM
- Access to the CUCM Web UI (<http://ipaddress>) and select Cisco Unified CM Administration from navigation drop down menu

Creating a SIP Profile in CUCM

To add a new SIP Profile in CUCM, login into the CUCM console, use the Device --- > Device settings --- > SIP Profile menu path in CUCM. Click on Add new and following are the settings, rest can be default:

Cisco Unified CM Administration
 For Cisco Unified Communications Solutions

System ▾ Call Routing ▾ Media Resources ▾ Advanced Features ▾ Device ▾ Application ▾ User Management ▾ Bulk Administration

SIP Profile Configuration

Save Delete Copy Reset Apply Config Add New

Name *	<input type="text" value="NTT SIP Profile"/>
Description	<input type="text"/>
Default MTP Telephony Event Payload Type *	<input type="text" value="101"/>
Early Offer for G.Clear Calls *	<input type="text" value="Disabled"/>
User-Agent and Server header information *	<input type="text" value="Send Unified CM Version Information as User-Age"/>
Version in User Agent and Server Header *	<input type="text" value="Major And Minor"/>
Dial String Interpretation *	<input type="text" value="Phone number consists of characters 0-9, *, #, an"/>
Confidential Access Level Headers *	<input type="text" value="Disabled"/>

Redirect by Application
 Disable Early Media on 180
 Outgoing T.38 INVITE include audio mline
 Use Fully Qualified Domain Name in SIP Requests
 Assured Services SIP conformance

SDP Information

SDP Session-level Bandwidth Modifier for Early Offer and Re-invites *	<input type="text" value="TIAS and AS"/>
SDP Transparency Profile	<input type="text" value="Pass all unknown SDP attributes"/>
Accept Audio Codec Preferences in Received Offer *	<input type="text" value="On"/>

SIP Profile Configuration

Save Delete Copy Reset Apply Config Add New

Trunk Specific Configuration

Reroute Incoming Request to new Trunk based on *	<input type="text" value="Never"/>
RSVP Over SIP *	<input type="text" value="Local RSVP"/>
Resource Priority Namespace List	<input type="text" value="< None >"/>

Fall back to local RSVP

SIP Rel1XX Options *	<input type="text" value="Send PRACK for all 1xx Messages"/>
Video Call Traffic Class *	<input type="text" value="Mixed"/>
Calling Line Identification Presentation *	<input type="text" value="Default"/>
Session Refresh Method *	<input type="text" value="Update"/>
Early Offer support for voice and video calls *	<input type="text" value="Mandatory (insert MTP if needed)"/>

Enable ANAT
 Deliver Conference Bridge Identifier
 Allow Passthrough of Configured Line Device Caller Information
 Reject Anonymous Incoming Calls
 Reject Anonymous Outgoing Calls
 Send ILS Learned Destination Route String

SIP Profile Configuration

Save ✖ Delete 📄 Copy 🔄 Reset 🔧 Apply Config ➕ Add New

Reject Anonymous Outgoing Calls
 Send ILS Learned Destination Route String

SIP OPTIONS Ping

 Enable OPTIONS Ping to monitor destination status for Trunks with Service Type "None (Default)"

Ping Interval for In-service and Partially In-service Trunks (seconds)*	60
Ping Interval for Out-of-service Trunks (seconds)*	120
Ping Retry Timer (milliseconds)*	500
Ping Retry Count*	6

SDP Information

 Send send-receive SDP in mid-call INVITE
 Allow Presentation Sharing using BFCP
 Allow iX Application Media
 Allow multiple codecs in answer SDP

Adding the E-SBC as a trunk in CUCM

The following process details the steps to add the SBC as a trunk in CUCM Web UI

1. On the CUCM administration console (UI), maneuver to **Device --- > Trunk**. Click on New
2. Select SIP Trunk from the Trunk Type drop down menu and protocol will also be SIP
3. Let default of none be selected on the Trunk service type
4. Following 2 screenshots are the other settings to be configured on the Trunk, all other parameters set to default

Trunk Configuration

Save
 Delete
 Reset
 Add New

Device Information

Product:	SIP Trunk
Device Protocol:	SIP
Trunk Service Type	None(Default)
Device Name*	<input type="text" value="NTT-Trunk"/>
Description	<input type="text"/>
Device Pool*	<input type="text" value="Default"/>
Common Device Configuration	<input type="text" value="< None >"/>
Call Classification*	<input type="text" value="Use System Default"/>
Media Resource Group List	<input type="text" value="< None >"/>
Location*	<input type="text" value="Hub_None"/>
AAR Group	<input type="text" value="< None >"/>
Tunneled Protocol*	<input type="text" value="None"/>
QSIG Variant*	<input type="text" value="No Changes"/>
ASN.1 ROSE OID Encoding*	<input type="text" value="No Changes"/>
Packet Capture Mode*	<input type="text" value="None"/>
Packet Capture Duration	<input type="text" value="0"/>

Trunk Configuration

Save
 Delete
 Reset
 Add New

SIP Information

Destination

Destination Address is an SRV

	Destination Address	Destination Address IPv6	Destination Port
1*	<input type="text" value="10.232.50.60"/>	<input type="text"/>	<input type="text" value="5060"/>

MTP Preferred Originating Codec*	<input type="text" value="711ulaw"/>
BLF Presence Group*	<input type="text" value="Standard Presence group"/>
SIP Trunk Security Profile*	<input type="text" value="Non Secure SIP Trunk Profile"/>
Rerouting Calling Search Space	<input type="text" value="< None >"/>
Out-Of-Dialog Refer Calling Search Space	<input type="text" value="< None >"/>
SUBSCRIBE Calling Search Space	<input type="text" value="< None >"/>
SIP Profile*	<input type="text" value="NTT SIP Profile"/> View Details
DTMF Signaling Method*	<input type="text" value="No Preference"/>

Creating a route Pattern in CUCM

Route pattern in CUCM take the form of regular expressions to define specific routes and give flexibility in network design for dialing outbound calls from CUCM users to the PSTN via the E-SBC. A route pattern comprises a string of digits (an address) and a set of associated digit manipulations that route calls to a route list or a gateway/trunk. In CUCM administration console, use the Call Routing --->Route/Hunt --->Route Pattern menu path to configure route patterns. Follow the fields in the screenshots below:

The screenshot shows the 'Route Pattern Configuration' page in Cisco Unified CM Administration. The 'Pattern Definition' section is active, showing the following fields and values:

- Route Pattern*: 8.@
- Route Partition: < None >
- Description: (empty)
- Numbering Plan*: NANP
- Route Filter: < None >
- MLPP Precedence*: Default
- Apply Call Blocking Percentage
- Resource Priority Namespace Network Domain: < None >
- Route Class*: Default
- Gateway/Route List*: ATT (with an [Edit](#) link)
- Route Option: Route this pattern, Block this pattern (No Error)

The screenshot shows the 'Route Pattern Configuration' page in Cisco Unified CM Administration, continuing from the previous section. The 'Called Party Transformations' and 'ISDN Network-Specific Facilities Information Element' sections are visible:

- Connected Name Presentation*: Default
- Called Party Transformations**
 - Discard Digits: PreDot
 - Called Party Transform Mask: (empty)
 - Prefix Digits (Outgoing Calls): (empty)
 - Called Party Number Type*: Cisco CallManager
 - Called Party Numbering Plan*: Cisco CallManager
- ISDN Network-Specific Facilities Information Element**
 - Network Service Protocol: -- Not Selected --
 - Carrier Identification Code: (empty)

Adding Devices/Phones and configuring Directory numbers

Cisco phones need to be added in CUCM by way of their MAC address and assigned to a specific user and then when powered on, they obtain an IP address in the CUCM topology with the subnet defined in CUCM administration console. Use the Device ---> Phone menu path to add new devices. One will need to define the template based on the device being configured, for example Cisco 9971 phone template as in the screenshots below. Also, some highlights of the configuration to add a user and configure a directory number (DN) to it in CUCM are shown below:

Cisco Unified CM Administration
For Cisco Unified Communications Solutions

Navigation: admin

System ▾ Call Routing ▾ Media Resources ▾ Advanced Features ▾ Device ▾ Application ▾ User Management ▾ Bulk Administration ▾ Help ▾

Phone Configuration Related Links: [Back To Find/List](#)

Save **X** Delete Copy Reset Apply Config Add New

----- Unassigned Associated Items -----

- 7 [Add a new SD](#)
- 8 All Calls
- 9 [Add a new BLF Directed Call Park](#)
- 10 Call Park
- 11 Call Pickup
- 12 CallBack
- 13 Group Call Pickup
- 14 Hunt Group Logout
- 15 [Intercom \[1\] - Add a new Intercom](#)
- 16 Malicious Call Identification
- 17 Meet Me Conference
- 18 Mobility

Device is Active
 Device is trusted

MAC Address* 580A209863BD

Description SEP580A209863BD-PurakATT

Device Pool* Default [View Details](#)

Common Device Configuration <None> [View Details](#)

Phone Button Template* Standard 9971 SIP

Softkey Template <None>

Common Phone Profile* Standard Common Phone Profile [View Details](#)

Calling Search Space <None>

AAR Calling Search Space <None>

Media Resource Group List

Cisco Unified CM Administration
For Cisco Unified Communications Solutions

Navigation: admin

System ▾ Call Routing ▾ Media Resources ▾ Advanced Features ▾ Device ▾ Application ▾ User Management ▾ Bulk Administration ▾ Help ▾

Phone Configuration Related Links: [Back To Find/List](#)

Save **X** Delete Copy Reset Apply Config Add New

Status

Status: Ready

Association

Modify Button Items

- 1 [Line \[1\] - 7322162709 \(no partition\)](#)
- 2 [Line \[2\] - Add a new DN](#)
- 3 [Add a new SD](#)
- 4 [Add a new SD](#)
- 5 [Add a new SD](#)
- 6 [Add a new SD](#)
- 7 [Add a new SD](#)
- 8 All Calls
- 9 [Add a new BLF Directed Call Park](#)
- 10 Call Park

----- Unassigned Associated Items -----

Phone Type

Product Type: Cisco 9971
Device Protocol: SIP

Real-time Device Status

Registration: Registered with Cisco Unified Communications Manager CUCM-Cisco
IPv4 Address: [10.232.50.79](#)
Active Load ID: sip9971.9-4-2-13
Inactive Load ID: sip9971.9-4-1-9
Download Status: None

Device Information

Device is Active
 Device is trusted

MAC Address* 580A209863BD

Description SEP580A209863BD-PurakATT

Creating & Assigning a Region and specifying use of G.729 codec in CUCM

ATT IP Flex Reach service certification covers AVPN transport and requires G.729 without annex b as choice of codec on the trunk side. CUCM 10.5 defaults to G.711 U law and therefore requires configuration to use G.729 codec. We can achieve this by the following steps:

- Specifying Audio codec preference
- Assigning it to a defined Region and
- finally assigning the region to the default Device Pool

In CUCM administration console, use the System --- >Region Information --- > Audio Codec Preference List and System --- >Region Information --- > Region menu path to configure these (Add new). Below screenshots provide an overview of the same:

Cisco Unified CM Administration
For Cisco Unified Communications Solutions

System ▾ Call Routing ▾ Media Resources ▾ Advanced Features ▾ Device ▾ Application ▾ User Management ▾ Bulk Administration ▾ Help ▾

Audio Codec Preference List Configuration

Save **X** Delete Copy Add New

Status: Ready

Audio Codec Preference List Information

Name*

Description*

Codecs in List*

- AMR-WB (7k-24k)
- AMR (5k-13k)
- MP4A-LATM 128k
- AAC-LD (MP4A Generic)
- MP4A-LATM 64k
- MP4A-LATM 56k
- L16 256k
- MP4A-LATM 48k
- ISAC 32k
- MP4A-LATM 32k
- G.722 64k
- G.722.1 32k
- G.722 56k
- G.722.1 24k
- G.722 48k
- MP4A-LATM 24k
- G.711 U-Law 64k
- G.711 A-Law 64k
- G.711 U-Law 56k
- G.711 A-Law 56k
- ILBC 16k
- G.728 16k

Cisco Unified CM Administration
For Cisco Unified Communications Solutions

System ▾ Call Routing ▾ Media Resources ▾ Advanced Features ▾ Device ▾ Application ▾ User Management ▾ Bulk Administration ▾ Help ▾

Region Configuration

Save **X** Delete Reset Apply Config Add New

Name*

Region Relationships

Region	Audio Codec Preference List	Maximum Audio Bit Rate	Maximum Session Bit Rate for Video Calls	Maximum
ATTflexreach	G729forCUCM	64 kbps (G.722, G.711)	384 kbps	
NOTE: Regions not displayed		Use System Default	Use System Default	Use System Default

Modify Relationship to other Regions

Regions	Audio Codec Preference List	Maximum Audio Bit Rate	Maximum Session Bit Rate for Video Calls
ATTflexreach			
Default			

Keep Current Setting Keep Current Setting

Next, assign the above defined region to the Device Pool to complete the configuration (System --- >Device Pool)

The screenshot shows the Cisco Unified CM Administration interface for Device Pool Configuration. The page title is "Cisco Unified CM Administration For Cisco Unified Communications Solutions". The navigation menu includes System, Call Routing, Media Resources, Advanced Features, Device, Application, User Management, and Bulk Administration. The main heading is "Device Pool Configuration". Below the heading are icons for Save, Delete, Copy, Reset, Apply Config, and Add New. The configuration is divided into three sections: "Device Pool Information", "Device Pool Settings", and "Roaming Sensitive Settings".

Device Pool Information	
Device Pool:	Default (27 members**)

Device Pool Settings	
Device Pool Name *	Default
Cisco Unified Communications Manager Group *	Default
Calling Search Space for Auto-registration	< None >
Adjunct CSS	< None >
Reverted Call Focus Priority	Default
Intercompany Media Services Enrolled Group	< None >

Roaming Sensitive Settings	
Date/Time Group *	CMLocal
Region *	ATTflexreach
Media Resource Group List	< None >

The CUCM is now ready to send/receive calls and establish SIP connectivity with the Oracle E-SBC.

Phase 2 – Configuring the Oracle Enterprise SBC

In this section we describe the steps for configuring an Oracle Enterprise SBC, formally known as an Acme Packet Net-Net Session Director (“SBC”), for use with Cisco Unified Call manager v10.5 in an NTT Communications SIP Trunk service.

In Scope

The following guide configuring the Oracle SBC assumes that this is a newly deployed device dedicated to a single customer. If a service provider currently has the SBC deployed and is adding CCM Server customers, then please see the ACLI Configuration Guide on http://docs.oracle.com/cd/E61547_01/index.html for a better understanding of the Command Line Interface (CLI).

Note that Oracle offers several models of SBC. This document covers the setup for the 4600 platform series running Net-Net OS ECZ7.5.0 or later. If instructions are needed for other Oracle SBC models, please contact your Oracle representative.

Out of Scope

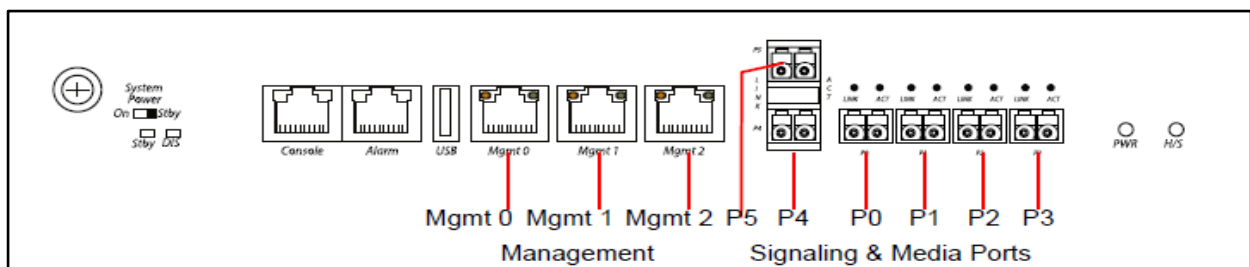
- Configuration of Network management including SNMP and RADIUS; and

What will you need

- Serial Console cross over cable with RJ-45 connector
- Terminal emulation application such as PuTTY or HyperTerm
- Passwords for the User and Superuser modes on the Oracle SBC
- IP address to be assigned to management interface (Wancom0) of the SBC - the Wancom0 management interface must be connected and configured to a management network separate from the service interfaces. Otherwise the SBC is subject to ARP overlap issues, loss of system access when the network is down, and compromising DDoS protection. Oracle does not support SBC configurations with management and media/service interfaces on the same subnet.
- IP address of Mediation Server external facing NIC
- IP addresses to be used for the SBC internal and external facing ports (Service Interfaces)
- IP address of the next hop gateway in the service provider network
- IP address of the enterprise DNS server

SBC- Getting Started

Once the Oracle SBC is racked and the power cable connected, you are ready to set up physical network connectivity. **Note: use the console port on the front of the SBC, not the one on the back.**



Plug the slot 0 port 0 (s0p0) interface into your outside (gateway facing) network and the slot 0 port 1 (s1p0) interface into your inside (CUCM server-facing) network. Once connected, perform you are ready to power on and perform the following steps.

All commands are in bold, such as **configure terminal**; parameters in bold red such as **oraclesbc1** are parameters which are specific to an individual deployment. **Note:** The ACLI is case sensitive.

Establish the serial connection and logging in the SBC

Confirm the SBC is powered off and connect one end of a straight-through Ethernet cable to the front console port (which is active by default) on the SBC and the other end to console adapter that ships with the SBC, connect the console adapter (a DB-9 adapter) to the DB-9 port on a workstation, running a terminal emulator application such as PuTTY. Start the terminal emulation application using the following settings:

- Baud Rate=115200
- Data Bits=8
- Parity=None
- Stop Bits=1
- Flow Control=None

Power on the SBC and confirm that you see the following output from the bootup sequence.

```
COM3 - PuTTY
Starting tEcmd...
Starting tSipd...
Starting tLtd...
Starting tH323d...
Starting tH248d...
Starting tBqfd...
Starting tSecured...
Starting tAuthd...
Starting tCerd...
Starting tIked...
Starting tauditd...
Starting tauditpusher...
Starting tSnmpd...
Start platform alarm...
Initializing /ramdrv Cleaner
Starting tLogCleaner task
Bringing up shell...
password secure mode is enabled
Admin Security is disabled
Starting SSH...
SSH Cli_init: allocated memory for 5 connections
accli: max telnet sessions: 5
Password: 0x21a059c8 (tAlarm): eth0: Link is up (1000Mb/s full duplex)
```

Enter the following commands to login to the SBC and move to the configuration mode. Note that the default SBC password is “**acme**” and the default super user password is “**packet**”.

```
Password: acme
oraclesbc1> enable
Password: packet
oraclesbc1# configure terminal
oraclesbc1 (configure)#
```

You are now in the global configuration mode.

Initial Configuration – Assigning the management Interface an IP address

To assign an IP address, one has to configure the bootparams on the SBC by going to

oraclesbc1#configure terminal --- >bootparams

- Once you type “bootparam” you have to use “carriage return” key to navigate down
- A reboot is required if changes are made to the existing bootparams

```
ACMESYSTEM(configure)# bootparam

'.' = clear field; '-' = go to previous field; q = quit

Boot File           : /boot/nnECZ750.bz
IP Address          : 192.65.79.44
VLAN                :
Netmask             : 255.255.255.224
Gateway             : 192.65.79.33
IPv6 Address        :
IPv6 Gateway        :
Host IP             : 0.0.0.0
FTP username        : vxftp
FTP password        : vxftp123
Flags               :
Target Name         : ACMESYSTEM
Console Device      : COM1
Console Baudrate    : 115200
Other               :

NOTE: These changed parameters will not go into effect until reboot.
Also, be aware that some boot parameters may also be changed through
PHY and Network Interface Configurations.
```

Configuring the SBC

The following section walks you through configuring the Oracle Communications Enterprise SBC configuration required to work with CUCM v10.5 and NTT's SIP Trunk service. In the configuration, the transport protocol used between the SBC and CUCM server is TCP and the SIP trunk is configured for UDP in this certification testing.

It is outside the scope of this document to include all the interoperability working information as it will differ in every deployment.

Surrogate registration

NTT requires the customer PBX to register in order to originate calls support authentication. Since CUCM cannot perform the registration, Oracle ESBC performs surrogate registrations on behalf of the PBX.

The configuration for surrogate registration is as follows

```
surrogate-agent
  register-host          ipvoice.jp
  register-user          +81XXXXXXXXXX
  description
  realm-id              CUCM
  state                 enabled
  customer-host
  customer-next-hop     ipvoice.jp
  register-contact-host  192.168.1.120
  register-contact-user  +81XXXXXXXXXX
  password              *****
  register-expires      3600
  replace-contact       disabled
  options
  route-to-registrar    enabled
  aor-count              1
  auth-user              user
  max-register-attempts  3
  register-retry-time    1800
  count-start           1
  register-mode          automatic
  triggered-inactivity-interval 30
  triggered-oos-response 503
```

The E-SBC provides authentication support required for the outgoing calls from CUCM. The auth-attribute configuration below is applied on the session-agent facing CUCM.

```

session-agent
  hostname                10.232.50.89
  transport-method        StaticTCP
  realm-id                 CUCM
  ping-method              OPTIONS
  ping-interval            30
  auth-attributes
    auth-realm             ipvoice.jp
    username                user
    password                *****
    in-dialog-methods      INVITE

```

SPLs required for NTT

As a part of the integration of the ESBC with NTT trunk, two SPLs, ocSurrogateRegister-1-4.pkg and ocNttMsgConverter-0-4.pkg were developed to include 5 features required to comply with the signaling requirements.

1. As a part of the surrogate registration, SBC is required to send a unique/random user-info portion in every REGISTER request that is sent to the NTT SIP trunk as well as outgoing INVITE messages for calls.
2. The ESBC is required to apply validity check to an incoming INVITE from the SIP trunk before sending out 100 TRYING and subsequent 1xx, 2xx messages to progress the call. It is expected that the incoming INVITE Request-URI user portion will contain the same randomized value that the E-SBC sent in the most recent REGISTER message to the trunk.
3. NTT regulation requires that the tag size of From/To headers in the SIP messages be under 32 bytes. The tags sent by CUCM in the originating SIP messages are large in size, approximately 51 bytes.
4. NTT specification also requires that the Rseq, Cseq, Session ID (in SDP) be under the value of 999900 and the SDP o line username character length be a maximum of 10 bytes. The E-SBC receives messages from Enterprise PBX – Cisco Unified Communications Manager with a large RSeq value in 18x messages which it forwards as is. Also, the SDP o line username is 19 bytes in length (generated by CUCM).
5. E-SBC is expected check RURI user portion of incoming CANCEL request for the AoR and compare it with the AoR specified in the Request-URI of the initial INVITE received.. If the value is different, E-SBC should respond with a 481 Call/Transaction Does Not Exist.

The SPL - **ocSurrogateRegister-1-6.pkg** was developed to implement the features 1 and 2. This SPL is enabled by configuring the spl-option **dyn-contact-start** on the sip-interface facing CUCM realm and **dyn-contact-method=randomseed** on the sip-interface facing the NTT trunk.

The SPL - **ocNttMsgConverter-0-3.pkg** was developed to implement the features 3, 4 and 5. This is enabled by configuring the spl-option **ocNttMsgConverterTagging=opposite** on the sip-interface facing CUCM realm and **ocNttMsgConverterTagging=enabled** on the sip-interface facing the NTT trunk.

Media policy

NTT requires that the TOS value for SIP and RTP be set to 5. The following media-policy is configured and applied on the realm-config towards NTT.

```

media-policy
  name                    NTT-Tos
  tos-settings
    media-type             message
    media-sub-type         sip
    tos-value              0x05
  tos-settings

```

media-type	audio
tos-value	0x05

Number translations

NTT requires the telephone numbers in the From and To headers to be in E164 format. Since CUCM does not send the numbers in E164 format, we configure a translation rule to add + to the uri-users of the From and To headers of the INVITEs going to NTT and apply it on the realm towards NTT.

translation-rules	
id	addforJP
type	add
add-string	+
session-translation	
id	addJP
rules-calling	addforJP
rules-called	addforJP
realm-config	
identifier	NTT-router
network-interfaces	s0p1:0
media-policy	NTT-Tos
out-translationid	addJP

A translation rule removeplus is configured to remove the plus from the telephone numbers in the From and To headers in the SIP messages being sent to CUCM and is applied on the realm towards CUCM.

translation-rules	
id	removeplus
type	delete
delete-string	+
session-translation	
id	DelJP
rules-calling	removeplus
rules-called	removeplus
realm-config	
identifier	CUCM
network-interfaces	s0p3:0
out-translationid	DelJP

SIP manipulations

CUCM and NTT SIP trunk carry their own SIP and SDP design – not always these implementation methods align, causing a lot of mis-match in SIP and SDP signaling and call flow. The ESBC helps resolve these issues with SIP manipulation feature

We have configured a sip-manipulation to modify the signaling according to NTT specifications. This manipulation, named Changecontact (which is a nested manipulation) is applied as an out-manipulationid on the sip-interface towards NTT.

Below is a list of all the manipulations referenced by the ChangeContact sip-manipulation.

1. NATting – configured for topology-hiding.
2. ModToport – NTT requires all the To headers to contain the port number (for e.g. sip:+815034252023@ipvoice.jp:7060). This manipulation adds the port to the To header if it does not exist.
3. PAlandRPI --- To delete the Remote-Party-ID and P-Asserted-Identity headers sent by CUCM.
4. AddSBCinfo – To replace the CUCM related information in the User-Agent header with the SBC image version.
5. removecallinfo – To delete the Call-Info header received from CUCM..
6. AddSupportedinReg – To add a “Supported: path” header in REGISTER messages.
7. ModSupportedoutboundINVITE – To replace the value of Supported header in INVITE with 100rel,timer
8. ForREGISTER – To add the required authentication details in the REGISTERs sent to NTT trunk.
9. RemDateCiscoGuideallowevents – Delete Date, Cisco-Guid and Allow-Events headers received from CUCM.
10. ModMaxforwards – To modify the Max-Forwards header value to 70 and adds the header if it is not present.
11. deltransportUDP – To remove the ‘transport’ uri-parameter from the Contact header.
12. Modcontactuserinresponses – To add + to the uri-user in Contact header.
13. ModAllowheader – Modifies the Allow header value in INVITE and UPDATE to include the methods, INVITE,BYE,CANCEL,ACK,PRACK,UPDATE and adds the Allow header if it is not present.
14. DelReasonheader – To delete the Reason header in BYE.
15. ModUPDATEmessage – To modify the Supported header in UPDATES to include only timer.
16. Addsupportedwithtimer – Adds a Supported: timer header to replies of INVITEs and UPDATES.
17. DelExpiresinINVITE – To delete the Expires header from the INVITE.
18. forsessionexpirestoNTT – To modify the value in the Session-Expires header to 180
19. anonymouscall - To remove the '+' from the uri-user in the RURI and To headers
20. striplines – To remove the unwanted lines from the SDP.

A sip-manipulation ToCUCM is configured as the out-manipulationid towards the CUCM. It refers to the following sip-manipulations.

1. NATting – configured for topology-hiding.
2. ModRURItocUCM – To replace the random contact in the uri-user of the RURI with that of the To header
3. ModSupportedtowardsCUCM – To modify the Supported header to include the Allow method.
4. PAlandRPI --- To delete the Remote-Party-ID and P-Asserted-Identity headers sent by CUCM.

A sip-manipulation Forsurragent is configured as the out-manipulationid towards the CUCM. It refers to the following sip-manipulations.

1. NATting – configured for topology-hiding.
2. ModRURItocUCM – To replace the random contact in the uri-user of the RURI with that of the To header
3. ModSupportedinINVITE – To delete the Supported headers if the value matches Geolocation and X-cisco-srtp-fallback

SBC Configuration

Following is the complete configuration of the SBC:

```
local-policy
  from-address      *
  to-address        *
  source-realm      CUCM
  policy-attribute
    next-hop        ipvoice.jp
    realm           NTT-router
    app-protocol    SIP
local-policy
  from-address      *
  to-address        *
  source-realm      NTT-router
  policy-attribute
    next-hop        10.232.50.89
    realm           CUCM
    action          replace-uri
    app-protocol    SIP
media-manager
media-policy
  name              NTT-Tos
  tos-settings
    media-type      message
    media-sub-type  sip
    tos-value       0x05
  tos-settings
    media-type      audio
    tos-value       0x05
media-profile
  name              PCMA
  payload-type      8
media-profile
  name              PCMU
  payload-type      0
network-interface
  name              s0p1
  description
  ip-address        192.168.1.120
  netmask           255.255.255.0
  gateway           192.68.1.1
  hip-ip-list       192.168.1.120
  icmp-address      192.168.1.120
  ssh-address
network-interface
  name              s0p3
  ip-address        10.232.50.60
  netmask           255.255.255.0
  gateway           10.232.50.89
  hip-ip-list       10.232.50.60
  ftp-address       10.232.50.60
  icmp-address      10.232.50.60
  ssh-address       10.232.50.60
phy-interface
  name              s0p0
  operation-type    Media
  admin-state       disabled
phy-interface
```

```

        name                s0p1
        operation-type      Media
        port                1
    phy-interface
        name                s0p2
        operation-type      Media
        port                2
        admin-state        disabled
    phy-interface
        name                s0p3
        operation-type      Media
        port                3
    realm-config
        identifier          CUCM
        network-interfaces s0p3:0
        out-translationid   DelJP
        spl-options         dyn-contact-start,ocNttMsgConverterTagging=opposite
    realm-config
        identifier          NTT-router
        network-interfaces s0p1:0
        media-policy        NTT-Tos
        out-translationid   addJP
        spl-options         ocNttMsgConverterTagging=enabled,dyn-contact-
method=randomseed
        codec-policy        offonlyPCMtoNTT
    session-agent
        hostname            10.232.50.89
        transport-method    StaticTCP
        realm-id            CUCM
        ping-method         OPTIONS
        ping-interval       30
        auth-attributes
            auth-realm      ipvoice.jp
            username        user
            password        *****
            in-dialog-methods INVITE
    session-agent
        hostname            ipvoice.jp
        ip-address          192.168.1.170
        port                7060
        realm-id            NTT-router
    session-translation
        id                  DelJP
        rules-calling       removeplus
        rules-called        removeplus
    session-translation
        id                  addJP
        rules-calling       addforJP
        rules-called        addforJP
    sip-config
        home-realm-id       CUCM
        registrar-domain    *
        registrar-host      *
        registrar-port      5060
        options              force-unregistration
                           max-udp-length=0
    sip-interface
        realm-id            CUCM
        sip-port
            address         10.232.50.60

```



```

transport-protocol          TCP
  allow-anonymous           agents-only
  registration-caching      enabled
  in-manipulationid        Forsurragent
  out-manipulationid        ToCUCM
sip-interface
  realm-id                  NTT-router
  sip-port
    address                  192.168.1.120
    allow-anonymous          registered
    out-manipulationid       Changecontact
sip-manipulation
  name                      AddSBCinfo
  header-rule
    name                    Addproductinfo
    header-name              User-Agent
    action                   add
    msg-type                 request
    methods                  REGISTER
    new-value                OracleE\-SBC/ECZ750
  header-rule
    name                    Moduseragentforcall
    header-name              User-Agent
    action                   manipulate
    comparison-type          pattern-rule
    msg-type                 request
    methods                  ACK, BYE, INVITE, PRACK, UPDATE
    element-rule
      name                   Modvalue
      type                   header-value
      action                 replace
      comparison-type        pattern-rule
      match-value            ^Cisco(.*)
      new-value              OracleE\-SBC/ECZ750
  header-rule
    name                    ChecServerheaderinbye
    header-name              Server
    action                   manipulate
    comparison-type          pattern-rule
    methods                  BYE, INVITE
    element-rule
      name                   Modvalue
      type                   header-value
      action                 replace
      comparison-type        pattern-rule
      match-value            ^Cisco(.*)
      new-value              OracleE\-SBC/ECZ750
  header-rule
    name                    delserverheader
    header-name              Server
    action                   delete
    methods                  BYE, INVITE, PRACK, UPDATE
sip-manipulation
  name                      AddSupportedinReg
  header-rule
    name                    Addtheheader
    header-name              Supported
    action                   add
    msg-type                 request
    methods                  REGISTER

```

	new-value	path
sip-manipulation		
name		Addsupportedwithtimer
header-rule		
name		Addsupported
header-name		Supported
action		add
msg-type		reply
methods		INVITE, UPDATE
new-value		timer
sip-manipulation		
name		Changecontact
header-rule		
name		forprivacy
header-name		From
action		sip-manip
new-value		NATting
header-rule		
name		forPAIandRPI
header-name		From
action		sip-manip
new-value		PAIandRPI
header-rule		
name		forUAinfo
header-name		From
action		sip-manip
new-value		AddSBCinfo
header-rule		
name		Toremovecallinfo
header-name		From
action		sip-manip
new-value		removecallinfo
header-rule		
name		forregsupport
header-name		From
action		sip-manip
new-value		AddSupportedinReg
header-rule		
name		outboundinvite
header-name		From
action		sip-manip
new-value		ModSupportedoutboundINVITE
header-rule		
name		regrule
header-name		From
action		sip-manip
new-value		ForREGISTER
header-rule		
name		fordateciscoguidalloweventsdel
header-name		From
action		sip-manip
new-value		RemDateCiscoGuideallowevents
header-rule		
name		formaxforwards
header-name		From
action		sip-manip
new-value		ModMaxforwards
header-rule		
name		fortransportudp
header-name		From

action	sip-manip
new-value	deltransportUDP
header-rule	
name	forplusinresponse
header-name	From
action	sip-manip
new-value	Modcontactuserinresponses
header-rule	
name	formodallowheader
header-name	From
action	sip-manip
new-value	ModAllowheader
header-rule	
name	forreasonheader
header-name	From
action	sip-manip
new-value	DelReasonheader
header-rule	
name	forupdatemessage
header-name	From
action	sip-manip
new-value	ModUPDATEmessage
header-rule	
name	fortimerinsupported
header-name	From
action	sip-manip
new-value	Addsupportedwithtimer
header-rule	
name	DeleteexpiresinINVITE
header-name	From
action	sip-manip
new-value	DelExpiresinINVITE
header-rule	
name	forSEtoNTT
header-name	From
action	sip-manip
new-value	forsessionexpirestoNTT
header-rule	
name	foranonymouscall
header-name	From
action	sip-manip
new-value	anonymouscall
header-rule	
name	remblines
header-name	From
action	sip-manip
new-value	stripblines
sip-manipulation	
name	DelExpiresinINVITE
header-rule	
name	delexpires
header-name	Expires
action	delete
msg-type	request
methods	INVITE
sip-manipulation	
name	DelReasonheader
header-rule	
name	delreason
header-name	Reason

action	delete
msg-type	request
methods	BYE
sip-manipulation	
name	ForREGISTER
header-rule	
name	Delroute
header-name	Route
action	delete
msg-type	request
methods	REGISTER
header-rule	
name	Delauthparams
header-name	Authorization
action	manipulate
msg-type	request
methods	REGISTER
element-rule	
name	storevalue
type	header-value
action	store
comparison-type	pattern-rule
match-value	(.+) (, auth-params=sha1-credential)
element-rule	
name	delparam
type	header-value
action	replace
comparison-type	pattern-rule
new-value	\$Delauthparams.\$storevalue.\$1
header-rule	
name	addContentlength
header-name	Content-Length
action	add
msg-type	request
methods	REGISTER
new-value	0
header-rule	
name	delexpires
header-name	Expires
action	delete
msg-type	request
methods	REGISTER
header-rule	
name	adduserphoneinFrom
header-name	From
action	manipulate
msg-type	request
methods	INVITE, REGISTER
element-rule	
name	adduserphone
parameter-name	user
type	uri-param
action	add
new-value	phone
header-rule	
name	adduserphoneinTo
header-name	To
action	manipulate
msg-type	request
methods	INVITE, REGISTER

```

        element-rule
            name                adduserphonto
            parameter-name       user
            type                 uri-param
            action               add
            new-value            phone
    header-rule
        name                    adduserphoneinRURIINVITE
        header-name             Request-URI
        action                  manipulate
        msg-type                request
        methods                 INVITE
        element-rule
            name                adduserequalphone
            parameter-name       user
            type                 uri-param
            action               add
            new-value            phone
    header-rule
        name                    Forinvitedelauthparams
        header-name             Proxy-Authorization
        action                  manipulate
        msg-type                request
        methods                 INVITE
        element-rule
            name                storethevalue
            type                 header-value
            action               store
            comparison-type      pattern-rule
            match-value          (.+)(, auth-params=sha1-credential)
        element-rule
            name                delparam
            type                 header-value
            action               replace
            comparison-type      pattern-rule
            new-value
$Forinvitedelauthparams.$storethevalue.$1
    header-rule
        name                    addopaqueinReg
        header-name             Authorization
        action                  manipulate
        comparison-type         pattern-rule
        msg-type                request
        methods                 REGISTER
        element-rule
            name                storeentireheader
            type                 header-value
            action               store
            comparison-type      pattern-rule
            match-value          (.+)(, algorithm=MD5)
        element-rule
            name                addopaqueparam
            parameter-name       opaque
            type                 header-value
            action               replace
            comparison-type      pattern-rule
            new-value
$addopaqueinReg.$storeentireheader.$1+$addopaqueinReg.$storeentireheader.$2+,+opaque="\\"
    header-rule
        name                    addopaqueinINVITE

```

```

header-name Proxy-authorization
action manipulate
msg-type request
methods INVITE
element-rule
  name Checkheader
  type header-value
  action store
  comparison-type pattern-rule
  match-value (.+)(, algorithm=MD5)
element-rule
  name addopaqueinheader
  type header-value
  action replace
  comparison-type pattern-rule
  new-value
$addopaqueinINVITE.$Checkheader.$1+$addopaqueinINVITE.$Checkheader.$2+,+opaque=\"\"
sip-manipulation
  name Forsurragent
  header-rule
    name forsupportedinINVITE
    header-name From
    action sip-manip
    new-value ModSupportedinINVITE
  header-rule
    name ChangeFrom
    header-name From
    action manipulate
    msg-type request
    methods INVITE
    element-rule
      name NTT_from_user
      parameter-name From
      type uri-user
      action replace
      new-value 815034252021
sip-manipulation
  name ModAllowheader
  header-rule
    name CheckAllowheader
    header-name Allow
    action manipulate
    methods INVITE, UPDATE
    element-rule
      name Storeheadervalue
      type header-value
      action store
      comparison-type pattern-rule
      match-value .*
    element-rule
      name Modallow
      type header-value
      action replace
      new-value INVITE, BYE, CANCEL, ACK, PRACK, UPDATE
  header-rule
    name Checkallowandifnotaddit
    header-name Allow
    action add
    comparison-type boolean
    msg-type request

```

```

        methods                UPDATE
        match-value             !$CheckAllowheader.$Storeheadervalue
        element-rule
            name                 addheadervalue
            type                 header-value
            action               add
            new-value            INVITE,BYE,CANCEL,ACK,PRACK,UPDATE
sip-manipulation
    name                       ModMaxforwards
    description                 Look for Max-Forwards header, change it to 70 and if
not present, add it
    header-rule
        name                   CheckMaxforwards
        header-name            Max-Forwards
        action                 manipulate
        methods                ACK,BYE,INVITE,PRACK,UPDATE
        element-rule
            name                 storevalue
            type                 header-value
            action               store
            comparison-type      pattern-rule
        element-rule
            name                 add70
            type                 header-value
            action               find-replace-all
            comparison-type      pattern-rule
            new-value            70
    header-rule
        name                   Addmaxforwardsifnotpresent
        header-name            Max-Forwards
        action                 add
        comparison-type        boolean
        msg-type                reply
        methods                ACK,BYE,INVITE,PRACK,UPDATE
        match-value            !$CheckMaxforwards.$storevalue
        element-rule
            name                 addvalue
            type                 header-value
            action               add
            new-value            70
sip-manipulation
    name                       ModRURIToCUCM
    header-rule
        name                   CheckToheader
        header-name            To
        action                 manipulate
        msg-type                request
        methods                INVITE
        element-rule
            name                 storeTouriuser
            type                 uri-user
            action               store
            comparison-type      pattern-rule
    header-rule
        name                   ModRURIUser
        header-name            Request-URI
        action                 manipulate
        msg-type                request
        methods                INVITE
        element-rule

```

```

        name                replaceuserinfo
        type                uri-user
        action              replace
        new-value           $CheckToheader.$storeTourius.$0
sip-manipulation
  name                    ModSupportedinINVITE
  header-rule
    name                  Checkciscosupported
    header-name           Supported
    action                delete
    comparison-type       pattern-rule
    methods               INVITE,UPDATE
    match-value           X-cisco-srtp(.*)
  header-rule
    name                  Checkgeosupported
    header-name           Supported
    action                delete
    methods               INVITE,UPDATE
    match-value           Geolocation
sip-manipulation
  name                    ModSupportedoutboundINVITE
  header-rule
    name                  CheckSupported
    header-name           Supported
    action                manipulate
    comparison-type       pattern-rule
    msg-type              request
    methods               INVITE
    element-rule
      name                Storevalue
      type                header-value
      action              store
      comparison-type     pattern-rule
    element-rule
      name                add100rel
      type                header-value
      action              find-replace-all
      comparison-type     pattern-rule
      new-value           100rel,timer
  header-rule
    name                  delsupportedin200
    header-name           Supported
    action                delete
    msg-type              reply
    methods               INVITE
    match-value           replaces
sip-manipulation
  name                    ModSupportedtowardsCUCM
  header-rule
    name                  CheckSupported
    header-name           Supported
    action                manipulate
    comparison-type       pattern-rule
    methods               INVITE
    element-rule
      name                Storevalue
      type                header-value
      action              store
      comparison-type     pattern-rule
    element-rule

```



```

        name                               isupdate
        type                               header-value
        action                             replace
        comparison-type                    pattern-rule
        new-value
$CheckSupported.$Storevalue.$0+,+UPDATE
sip-manipulation
  name                                     ModToheader
  header-rule
    name                                   CheckToheader
    header-name                           To
    action                                manipulate
    methods                               ACK,UPDATE
    element-rule
      name                                 Storevalue
      type                                 uri-port
      action                               store
      comparison-type                    pattern-rule
      match-value                         .*
  header-rule
    name                                   CheckdoubleportsinTo
    header-name                           To
    action                                manipulate
    comparison-type                      boolean
    methods                               ACK,UPDATE
    match-value                           $CheckToheader.$Storevalue
    element-rule
      name                                 ChangeTovalue
      type                                 uri-port
      action                               replace
      new-value                           7060
sip-manipulation
  name                                     ModToport
  header-rule
    name                                   CheckToport
    header-name                           To
    action                                manipulate
    element-rule
      name                                 Storeport
      type                                 uri-port
      action                               store
      match-value                         7060
  header-rule
    name                                   CheckdoubleportsinTo
    header-name                           To
    action                                manipulate
    comparison-type                      boolean
    match-value                           !$CheckToport.$Storeport
    element-rule
      name                                 ChangeToval
      type                                 uri-port
      action                               add
      new-value                           7060
sip-manipulation
  name                                     ModUPDATEmessage
  header-rule
    name                                   ModSupportedheader
    header-name                           Supported
    action                                manipulate
    comparison-type                      pattern-rule

```

```

msg-type request
methods UPDATE
element-rule
  name keeptimeronly
  type header-value
  action replace
  comparison-type pattern-rule
  new-value timer
sip-manipulation
  name Modcontactuserinresponses
  header-rule
    name Modtheuser
    header-name Contact
    action manipulate
    msg-type reply
    methods INVITE,UPDATE
    element-rule
      name Checkuser
      type uri-user
      action store
      comparison-type pattern-rule
    element-rule
      name addplussign
      type uri-user
      action replace
      new-value \++$Modtheuser.$Checkuser.$0
sip-manipulation
  name NATting
  header-rule
    name From
    header-name From
    action manipulate
    element-rule
      name From_header
      type uri-host
      action replace
      new-value ipvoice.jp
  header-rule
    name To
    header-name To
    action manipulate
    element-rule
      name To
      type uri-host
      action replace
      new-value ipvoice.jp
    element-rule
      name Toport
      type uri-port
      action sip-manip
      new-value ModToport
sip-manipulation
  name PAIandRPI
  header-rule
    name delRPI
    header-name Remote-Party-ID
    action delete
    methods INVITE,UPDATE
  header-rule
    name delPAI

```

header-name	P-Asserted-Identity
action	delete
methods	BYE, INVITE, UPDATE
sip-manipulation	
name	RemDateCiscoGuideallowevents
header-rule	
name	delDate
header-name	Date
action	delete
methods	ACK, BYE, INVITE, PRACK, UPDATE
header-rule	
name	delCiscoGuid
header-name	Cisco-Guid
action	delete
methods	ACK, BYE, INVITE, PRACK, UPDATE
header-rule	
name	delAllowevents
header-name	Allow-Events
action	delete
methods	ACK, BYE, INVITE, PRACK, UPDATE
sip-manipulation	
name	StripSDPfromPRACK
header-rule	
name	delSDPfromPRACK
header-name	Content-Type
action	manipulate
msg-type	request
methods	PRACK
element-rule	
name	delpracksdp
parameter-name	application/sdp
type	mime
action	delete-element
header-rule	
name	ChangeCLlength
header-name	Content-Length
action	manipulate
comparison-type	boolean
msg-type	request
methods	PRACK
match-value	\$delSDPfromPRACK.\$delpracksdp
element-rule	
name	Changelegnthto0
type	header-value
action	replace
new-value	0
sip-manipulation	
name	ToCUCM
header-rule	
name	ForNAT_IP
header-name	From
action	sip-manip
new-value	Topohiding
header-rule	
name	forRURI
header-name	From
action	sip-manip
new-value	ModRURItocUCM
header-rule	
name	ForupdatetocUCM

```

        header-name      From
        action           sip-manip
        new-value        ModSupportedtowardsCUCM
sip-manipulation
  name                  Topohiding
  header-rule
    name                From
    header-name         From
    action              manipulate
    element-rule
      name              From_header
      type              uri-host
      action             replace
      new-value         $LOCAL_IP
  header-rule
    name                To
    header-name         To
    action              manipulate
    element-rule
      name              To
      type              uri-host
      action             replace
      new-value         $REMOTE_IP
sip-manipulation
  name                  anonymouscall
  header-rule
    name                changeRURI
    header-name         Request-URI
    action              manipulate
    msg-type            request
    methods             INVITE
    element-rule
      name              storeuser
      type              uri-user
      action             store
      comparison-type   pattern-rule
      match-value       ^\+184 (.*)$
    element-rule
      name              striptheplus
      type              uri-user
      action             replace
      comparison-type   boolean
      match-value       $changeRURI.$storeuser
      new-value         $ORIGINAL-^"+
  header-rule
    name                addphonecontext
    header-name         Request-URI
    action              manipulate
    comparison-type     boolean
    msg-type            request
    methods             INVITE
    match-value         $changeRURI.$storeuser.$0
    element-rule
      name              addtheparam
      parameter-name    phone-context
      type              uri-user-param
      action             add
      new-value         \+81
  header-rule
    name                ModToheader

```

```

        header-name      To
        action            manipulate
        comparison-type   pattern-rule
        msg-type          request
        methods           INVITE
        element-rule
            name          storetheuser
            type          uri-user
            action        store
            comparison-type pattern-rule
            match-value   ^\+184(.*$)
        element-rule
            name          Striptheplusfromuriuser
            type          uri-user
            action        replace
            comparison-type boolean
            match-value   $ModToheader.$storetheuser
            new-value     $ORIGINAL-^"+
    header-rule
        name              addphonecontextinTo
        header-name      To
        action            manipulate
        comparison-type   boolean
        msg-type          request
        methods           INVITE
        match-value      $ModToheader.$storetheuser.$0
        element-rule
            name          addpc
            parameter-name phone-context
            type          uri-user-param
            action        add
            new-value     \+81
sip-manipulation
    name                deltransportUDP
    header-rule
        name            Remtransportudp
        header-name     Contact
        action          manipulate
        methods         INVITE,UPDATE
        element-rule
            name          delparam
            parameter-name transport
            type          uri-param
            action        delete-element
sip-manipulation
    name                drop100
    header-rule
        name            checkRURI
        header-name     Request-URI
        action          manipulate
        msg-type        reply
        methods         INVITE
        element-rule
            name          storemsgcode
            type          status-code
            action        store
            comparison-type pattern-rule
            match-value   100
        element-rule
            name          dropthemessage

```

```

        type                                header-value
        action                               reject
        match-value                          $checkRURI.$storemsgcode.$0
sip-manipulation
  name                                       forsessionexpirestonTT
  header-rule
    name                                     adduacforSE
    header-name                             Session-Expires
    action                                  manipulate
    comparison-type                         pattern-rule
    msg-type                                request
    methods                                 INVITE
    element-rule
      name                                   storesevalue
      type                                  header-value
      action                                 store
      comparison-type                       pattern-rule
      match-value                           (.*)
    element-rule
      name                                   addrefresheruac
      type                                  header-value
      action                                 replace
      comparison-type                       pattern-rule
      new-value                              180+;+refresher=uac
sip-manipulation
  name                                       prackwork
  header-rule
    name                                     delsupported
    header-name                             Supported
    action                                  delete
    msg-type                                request
    methods                                 INVITE
  header-rule
    name                                     addrequirereinINVITE
    header-name                             Require
    action                                  add
    msg-type                                request
    methods                                 INVITE
    new-value                               100rel
sip-manipulation
  name                                       remove2ports
  header-rule
    name                                     Removedoubleport
    header-name                             To
    action                                  manipulate
    methods                                 ACK, PRACK, UPDATE
    element-rule
      name                                   To
      type                                  uri-host
      action                                 replace
      new-value                              ipvoice.jp
sip-manipulation
  name                                       removecallinfo
  header-rule
    name                                     Stripcallinfo
    header-name                             Call-Info
    action                                  delete
    methods                                 INVITE, UPDATE
sip-manipulation
  name                                       stripblines

```

```

header-rule
  name                               blinefix
  header-name                         Content-Type
  action                               manipulate
  element-rule
    name                               removeb1
    parameter-name                     application/sdp
    type                               mime
    action                             find-replace-all
    comparison-type                    pattern-rule
    match-value                        b=TIAS:64000\r\n
  element-rule
    name                               removeb2
    parameter-name                     application/sdp
    type                               mime
    action                             find-replace-all
    comparison-type                    pattern-rule
    match-value                        b=AS:64\r\n
  element-rule
    name                               removemaxptime
    parameter-name                     application/sdp
    type                               mime
    action                             find-replace-all
    comparison-type                    pattern-rule
    match-value                        a=maxptime:20\r\n
sip-monitoring
  match-any-filter                     enabled
spl-config
  plugins
    name                               ocSurrogateRegister-1-5.pkg
  plugins
    name                               ocNttMsgConverter-0-3.pkg
steering-pool
  ip-address                           192.168.1.120
  start-port                           41000
  end-port                              45000
  realm-id                              NTT-router
steering-pool
  ip-address                           10.232.50.60
  start-port                           35000
  end-port                              37000
  realm-id                              CUCM
surrogate-agent
  register-host                         ipvoice.jp
  register-user                         +81XXXXXXXXXX
  realm-id                              CUCM
  customer-next-hop                     ipvoice.jp
  register-contact-host                 192.168.1.120
  register-contact-user                 +81XXXXXXXXXX
  password                              *****
  register-expires                      3600
  auth-user                             user
  register-retry-time                   1800
system-config
  process-log-level                     DEBUG
  default-gateway                       192.168.1.115
translation-rules
  id                                    addforJP
  type                                  add

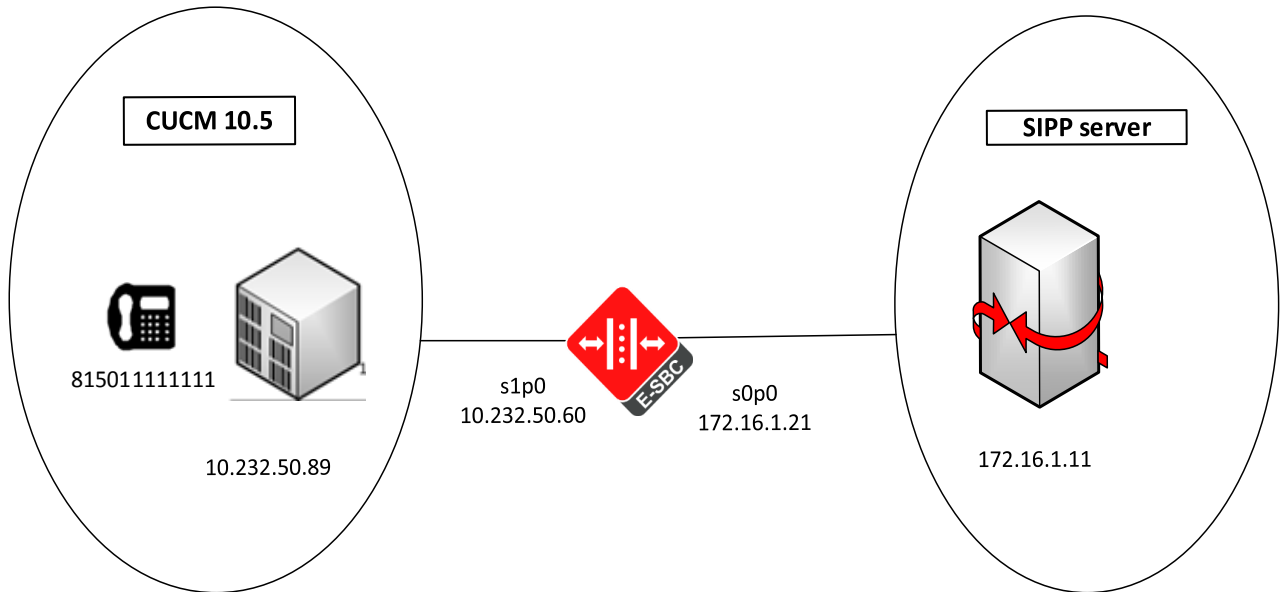
```

add-string	+
translation-rules	
id	removeplus
type	delete
delete-string	+
web-server-config	

SIPP based testing

Architecture

In addition to the trunk based test cases, the testing also included an exhaustive list of test cases that utilized SIPP scripts. These tests simulate corner case behavior either by CUCM or the NTT SIP trunk. The configuration for this part of testing is slightly different from the trunk based test configuration. The architecture diagram below shows the setup utilized for SIPP testing.



The ESBC configurations have been included in this guide. The configuration consists of sip-manipulations that are required for specific test cases.

For any questions regarding the configuration or the test cases, please contact your Oracle sales representative.

SBC Configuration

```
codec-policy
  name TC1283
  allow-codecs PCMU
codec-policy
  name toCUCM
  allow-codecs *
  order-codecs PCMU *
local-policy
  from-address *
  to-address *
  source-realm CUCM
  policy-attribute
    next-hop ntt.com
    realm NTT-router
    app-protocol SIP
local-policy
  from-address *
  to-address *
  source-realm NTT-router
  policy-attribute
    next-hop 10.232.50.89
    realm CUCM
    app-protocol SIP
media-manager
media-policy
  name NTT-Tos
  tos-settings
    media-type message
    media-sub-type sip
    tos-value 0x05
media-profile
  name PCMA
  payload-type 8
media-profile
  name PCMU
  payload-type 0
network-interface
  name s0p1
  description to SIPp
  ip-address 172.16.1.21
  netmask 255.255.255.0
  gateway 172.16.1.15
  hip-ip-list 172.16.1.21
  icmp-address 172.16.1.21
  ssh-address 172.16.1.21
network-interface
  name s0p3
  ip-address 10.232.50.60
  netmask 255.255.255.0
  gateway 10.232.50.89
  hip-ip-list 10.232.50.60
  ftp-address 10.232.50.60
  icmp-address 10.232.50.60
  ssh-address 10.232.50.60
phy-interface
  name s0p1
  operation-type Media
```

```

port 1
phy-interface
  name s0p3
  operation-type Media
  port 3
realm-config
  identifier CUCM
  network-interfaces s0p3:0
  out-translationid DelJP
  spl-options dyn-contact-start,ocNttMsgConverterTagging=opposite
  codec-policy toCUCM
realm-config
  identifier NTT-router
  network-interfaces s0p1:0
  media-policy NTT-Tos
  out-translationid add81forJP
  spl-options ocNttMsgConverterTagging=enabled,dyn-contact-
method=randomseed
session-agent
  hostname 10.232.50.89
  transport-method StaticTCP
  realm-id CUCM
  auth-attributes
    auth-realm ntt.com
    username user
    password *****
    in-dialog-methods INVITE
session-agent
  hostname ntt.com
  ip-address 172.16.1.11
  port 7060
  realm-id NTT-router
session-translation
  id DelJP
  rules-calling remove81
  rules-called remove81
session-translation
  id add81forJP
  rules-calling addforJP
  rules-called addforJP
sip-config
  home-realm-id CUCM
  registrar-domain *
  registrar-host *
  registrar-port 5060
  options inmanip-before-validate
  max-udp-length=0
sip-interface
  realm-id CUCM
  sip-port
    address 10.232.50.60
    transport-protocol TCP
    allow-anonymous agents-only
  registration-caching enabled
  in-manipulationid Forsurragent
  out-manipulationid ToCUCM
sip-interface
  realm-id NTT-router
  sip-port
    address 172.16.1.21

```

allow-anonymous	registered
registration-caching	enabled
options	droppresence=698
stop-recurse	401,407,580
in-manipulationid	inboundNTT
out-manipulationid	Changecontact
sip-manipulation	
name	AddSBCinfo
header-rule	
name	Addproductinfo
header-name	User-Agent
action	add
msg-type	request
methods	REGISTER
new-value	OracleE\ -SBC/ECZ750
header-rule	
name	Moduseragentforall
header-name	User-Agent
action	manipulate
comparison-type	pattern-rule
msg-type	request
methods	ACK, BYE, INVITE, PRACK, UPDATE
element-rule	
name	Modvalue
type	header-value
action	replace
comparison-type	pattern-rule
match-value	^Cisco(.*)
new-value	OracleE\ -SBC/ECZ750
header-rule	
name	ChecServerheaderinbye
header-name	Server
action	manipulate
comparison-type	pattern-rule
methods	BYE, INVITE
element-rule	
name	Modvalue
type	header-value
action	replace
comparison-type	pattern-rule
match-value	^Cisco(.*)
new-value	OracleE\ -SBC/ECZ730m1p1
header-rule	
name	delserverheader
header-name	Server
action	delete
methods	BYE, INVITE, PRACK, UPDATE
sip-manipulation	
name	AddSupportedinReg
header-rule	
name	Addtheheader
header-name	Supported
action	add
msg-type	request
methods	REGISTER
new-value	path
sip-manipulation	
name	Addsupportedwithtimer
header-rule	
name	Addsupported

header-name	Supported
action	add
msg-type	reply
methods	INVITE, UPDATE
new-value	timer
sip-manipulation	
name	Change100to699
header-rule	
name	check100
header-name	@status-line
action	store
comparison-type	pattern-rule
element-rule	
name	is100
type	status-code
action	store
comparison-type	pattern-rule
match-value	100
header-rule	
name	change100to699
header-name	@status-line
action	manipulate
comparison-type	boolean
msg-type	reply
methods	INVITE
match-value	check100.is100
element-rule	
name	modstatuscode
type	status-code
action	replace
new-value	699
element-rule	
name	delreasonphrase
type	reason-phrase
action	find-replace-all
sip-manipulation	
name	Changecontact
header-rule	
name	forprivacy
header-name	From
action	sip-manip
new-value	NATting
header-rule	
name	forPAIandRPI
header-name	From
action	sip-manip
new-value	PAIandRPI
header-rule	
name	forUAinfo
header-name	From
action	sip-manip
new-value	AddSBCinfo
header-rule	
name	Toremovecallinfo
header-name	From
action	sip-manip
new-value	removecallinfo
header-rule	
name	forregsupport
header-name	From

action	sip-manip
new-value	AddSupportedinReg
header-rule	
name	outboundinvite
header-name	From
action	sip-manip
new-value	ModSupportedoutboundINVITE
header-rule	
name	regrule
header-name	From
action	sip-manip
new-value	ForREGISTER
header-rule	
name	fordateciscoguidalloweventsdel
header-name	From
action	sip-manip
new-value	RemDateCiscoGuideallowevents
header-rule	
name	formaxforwards
header-name	From
action	sip-manip
new-value	ModMaxforwards
header-rule	
name	fortransportudp
header-name	From
action	sip-manip
new-value	deltransportUDP
header-rule	
name	forplusinresponse
header-name	From
action	sip-manip
new-value	Modcontactuserinresponses
header-rule	
name	formodallowheader
header-name	From
action	sip-manip
new-value	ModAllowheader
header-rule	
name	forreasonheader
header-name	From
action	sip-manip
new-value	DelReasonheader
header-rule	
name	forupdatemessage
header-name	From
action	sip-manip
new-value	ModUPDATEmessage
header-rule	
name	fortimerinsupported
header-name	From
action	sip-manip
new-value	Addsupportedwithtimer
header-rule	
name	DeleteexpiresinINVITE
header-name	From
action	sip-manip
new-value	DelExpiresinINVITE
header-rule	
name	forSEtoNTT
header-name	From

action	sip-manip
new-value	for session expires to NTT
header-rule	
name	for anonymous call
header-name	From
action	sip-manip
new-value	anonymous call
sip-manipulation	
name	DelExpiresinINVITE
header-rule	
name	delexpires
header-name	Expires
action	delete
msg-type	request
methods	INVITE
sip-manipulation	
name	DelReasonheader
header-rule	
name	delreason
header-name	Reason
action	delete
msg-type	request
methods	BYE
sip-manipulation	
name	ForREGISTER
header-rule	
name	Delroute
header-name	Route
action	delete
msg-type	request
methods	REGISTER
header-rule	
name	Delauthparams
header-name	Authorization
action	manipulate
msg-type	request
methods	REGISTER
element-rule	
name	storevalue
type	header-value
action	store
comparison-type	pattern-rule
match-value	(.+) (, auth-params=sha1-credential)
element-rule	
name	delparam
type	header-value
action	replace
comparison-type	pattern-rule
new-value	\$Delauthparams.\$storevalue.\$1
header-rule	
name	addContentlength
header-name	Content-Length
action	add
msg-type	request
methods	REGISTER
new-value	0
header-rule	
name	delexpires
header-name	Expires
action	delete

```

msg-type request
methods REGISTER
header-rule
  name adduserphoneinFrom
  header-name From
  action manipulate
  msg-type request
  methods INVITE,REGISTER
  element-rule
    name adduserphone
    parameter-name user
    type uri-param
    action add
    new-value phone
header-rule
  name adduserphoneinTo
  header-name To
  action manipulate
  msg-type request
  methods INVITE,REGISTER
  element-rule
    name adduserphonto
    parameter-name user
    type uri-param
    action add
    new-value phone
header-rule
  name adduserphoneinRURIINVITE
  header-name Request-URI
  action manipulate
  msg-type request
  methods INVITE
  element-rule
    name adduserequalphone
    parameter-name user
    type uri-param
    action add
    new-value phone
header-rule
  name Forinvitedelauthparams
  header-name Proxy-Authorization
  action manipulate
  msg-type request
  methods INVITE
  element-rule
    name storethevalue
    type header-value
    action store
    comparison-type pattern-rule
    match-value (.+)(, auth-params=sha1-credential)
  element-rule
    name delparam
    type header-value
    action replace
    comparison-type pattern-rule
    new-value
$Forinvitedelauthparams.$storethevalue.$1
header-rule
  name addopaqueinReg
  header-name Authorization

```



```

        action                manipulate
        comparison-type       pattern-rule
        msg-type               request
        methods                REGISTER
        element-rule
            name                storeentireheader
            type                header-value
            action              store
            comparison-type     pattern-rule
            match-value         (.+)(, algorithm=MD5)
        element-rule
            name                addopaqueparam
            parameter-name      opaque
            type                header-value
            action              replace
            comparison-type     pattern-rule
            new-value
$addopaqueinReg.$storeentireheader.$1+$addopaqueinReg.$storeentireheader.$2+,+opaque="\\"
        header-rule
            name                addopaqueinINVITE
            header-name         Proxy-authorization
            action              manipulate
            msg-type            request
            methods             INVITE
        element-rule
            name                Checkheader
            type                header-value
            action              store
            comparison-type     pattern-rule
            match-value         (.+)(, algorithm=MD5)
        element-rule
            name                addopaqueinheader
            type                header-value
            action              replace
            comparison-type     pattern-rule
            new-value
$addopaqueinINVITE.$Checkheader.$1+$addopaqueinINVITE.$Checkheader.$2+,+opaque="\\"
sip-manipulation
    name                        Forsurragent
    header-rule
        name                    forsUPPORTEDinINVITE
        header-name             From
        action                  sip-manip
        new-value                ModSupportedinINVITE
    header-rule
        name                    ChangeFrom
        header-name             From
        action                  manipulate
        msg-type                request
        methods                 INVITE
        element-rule
            name                  NTT_from_user
            parameter-name        From
            type                  uri-user
            action                replace
            new-value             815034252021
sip-manipulation
    name                        ModAllowheader
    header-rule
        name                    CheckAllowheader

```

header-name	Allow
action	manipulate
methods	INVITE,UPDATE
element-rule	
name	Storeheadervalue
type	header-value
action	store
comparison-type	pattern-rule
match-value	.*
element-rule	
name	Modallow
type	header-value
action	replace
new-value	INVITE,BYE,CANCEL,ACK,PRACK,UPDATE
header-rule	
name	Checkallowandifnotaddit
header-name	Allow
action	add
comparison-type	boolean
msg-type	request
methods	UPDATE
match-value	!\$CheckAllowheader.\$Storeheadervalue
element-rule	
name	addheadervalue
type	header-value
action	add
new-value	INVITE,BYE,CANCEL,ACK,PRACK,UPDATE
sip-manipulation	
name	ModMaxforwards
description	Look for Max-Forwards header, change it to 70 and if
not present, add it	
header-rule	
name	CheckMaxforwards
header-name	Max-Forwards
action	manipulate
methods	ACK,BYE,INVITE,PRACK,UPDATE
element-rule	
name	storevalue
type	header-value
action	store
comparison-type	pattern-rule
element-rule	
name	add70
type	header-value
action	find-replace-all
comparison-type	pattern-rule
new-value	70
header-rule	
name	Addmaxforwardsifnotpresent
header-name	Max-Forwards
action	add
comparison-type	boolean
msg-type	reply
methods	ACK,BYE,INVITE,PRACK,UPDATE
match-value	!\$CheckMaxforwards.\$storevalue
element-rule	
name	addvalue
type	header-value
action	add
new-value	70

```

sip-manipulation
  name                               ModRURItocUCM
  header-rule
    name                               CheckToheader
    header-name                         To
    action                               manipulate
    msg-type                             request
    methods                              INVITE
    element-rule
      name                               storeTouriuser
      type                               uri-user
      action                             store
      comparison-type                   pattern-rule
  header-rule
    name                               ModRURIuser
    header-name                         Request-URI
    action                               manipulate
    msg-type                             request
    methods                              INVITE
    element-rule
      name                               replaceuserinfo
      type                               uri-user
      action                             replace
      new-value                          $CheckToheader.$storeTouriuser.$0
sip-manipulation
  name                               ModSupportedinINVITE
  header-rule
    name                               Checkciscosupported
    header-name                         Supported
    action                               delete
    comparison-type                   pattern-rule
    methods                              INVITE,UPDATE
    match-value                        X-cisco-srtp(.*)
  header-rule
    name                               Checkgeosupported
    header-name                         Supported
    action                               delete
    methods                              INVITE,UPDATE
    match-value                        Geolocation
sip-manipulation
  name                               ModSupportedoutboundINVITE
  header-rule
    name                               CheckSupported
    header-name                         Supported
    action                               manipulate
    comparison-type                   pattern-rule
    msg-type                             request
    methods                              INVITE
    element-rule
      name                               Storevalue
      type                               header-value
      action                             store
      comparison-type                   pattern-rule
    element-rule
      name                               add100rel
      type                               header-value
      action                             find-replace-all
      comparison-type                   pattern-rule
      new-value                          100rel,timer
  header-rule

```

```

        name                                delsupportedin200
        header-name                          Supported
        action                                delete
        msg-type                              reply
        methods                              INVITE
        match-value                          replaces
sip-manipulation
  name                                       ModSupportedtowardsCUCM
  header-rule
    name                                     CheckSupported
    header-name                             Supported
    action                                   manipulate
    comparison-type                         pattern-rule
    methods                                  INVITE
    element-rule
      name                                   Storevalue
      type                                   header-value
      action                                   store
      comparison-type                       pattern-rule
    element-rule
      name                                   isupdate
      type                                   header-value
      action                                   replace
      comparison-type                       pattern-rule
      new-value
$CheckSupported.$Storevalue.$0+,+UPDATE
sip-manipulation
  name                                       ModToport
  header-rule
    name                                     CheckToport
    header-name                             To
    action                                   manipulate
    comparison-type                         pattern-rule
    match-value                             sip:*
    element-rule
      name                                   Storeport
      type                                   uri-port
      action                                   store
      match-value                           7060
  header-rule
    name                                     CheckdoubleportsinTo
    header-name                             To
    action                                   manipulate
    comparison-type                         boolean
    match-value                             !$CheckToport.$Storeport
    element-rule
      name                                   ChangeToval
      type                                   uri-port
      action                                   add
      new-value                             7060
sip-manipulation
  name                                       ModTotal
  header-rule
    name                                     CheckTotal
    header-name                             To
    action                                   manipulate
    comparison-type                         pattern-rule
    match-value                             tel:*
    element-rule
      name                                   delport

```

```

        type uri-port
        action replace
        match-value 7060
sip-manipulation
  name ModUPDATEmessage
  header-rule
    name ModSupportedheader
    header-name Supported
    action manipulate
    comparison-type pattern-rule
    msg-type request
    methods UPDATE
    element-rule
      name keeptimeronly
      type header-value
      action replace
      comparison-type pattern-rule
      new-value timer
sip-manipulation
  name Modcontactuserinresponses
  header-rule
    name Modtheuser
    header-name Contact
    action manipulate
    msg-type reply
    methods INVITE, UPDATE
    element-rule
      name Checkuser
      type uri-user
      action store
      comparison-type pattern-rule
    element-rule
      name addplussign
      type uri-user
      action replace
      new-value \++$Modtheuser.$Checkuser.$0
sip-manipulation
  name NATting
  header-rule
    name From
    header-name From
    action manipulate
    element-rule
      name From_header
      type uri-host
      action replace
      new-value ntt.com
  header-rule
    name To
    header-name To
    action manipulate
    comparison-type pattern-rule
    match-value sip:*
    element-rule
      name To
      type uri-host
      action replace
      new-value ntt.com
  element-rule
    name Toport

```

```

                                type                                uri-port
                                action                             sip-manip
                                new-value                         ModToport
sip-manipulation
  name                            PAIandRPI
  header-rule
    name                            delRPI
    header-name                      Remote-Party-ID
    action                            delete
    methods                          INVITE, UPDATE
  header-rule
    name                            delPAI
    header-name                      P-Asserted-Identity
    action                            delete
    methods                          BYE, INVITE, UPDATE
sip-manipulation
  name                            RemDateCiscoGuideallowevents
  header-rule
    name                            delDate
    header-name                      Date
    action                            delete
    methods                          ACK, BYE, INVITE, PRACK, UPDATE
  header-rule
    name                            delCiscoGuid
    header-name                      Cisco-Guid
    action                            delete
    methods                          ACK, BYE, INVITE, PRACK, UPDATE
  header-rule
    name                            delAllowevents
    header-name                      Allow-Events
    action                            delete
    methods                          ACK, BYE, INVITE, PRACK, UPDATE
sip-manipulation
  name                            TC1282
  mime-sdp-rule
    name                            rejip6
    msg-type                        request
    action                            manipulate
  sdp-session-rule
    name                            test_ip6
    action                            manipulate
  sdp-line-rule
    name                            checkip6
    type                            o
    action                            reject
    comparison-type                 pattern-rule
    match-value                      ^([0-9]{10}) ([0-9]{10}) ([0-
9]{10}) (IN IP6 .*)$
    new-value                         "403:Not Acceptable Protocol"
sip-manipulation
  name                            TC1283
  mime-sdp-rule
    name                            rejcodec
    msg-type                        request
    action                            manipulate
  sdp-media-rule
    name                            test_m
    media-type                       audio
    action                            manipulate
  sdp-line-rule

```

```

name change_payload
type m
action reject
comparison-type pattern-rule
match-value ^(audio) ( [0-9]{4,5}) ( RTP/AVP
9 15 18 4)$
new-value "403:Codecs Not Allowed"

sip-manipulation
name TC1284
mime-sdp-rule
name rejudp
msg-type request
action manipulate
sdp-media-rule
name test_udp
media-type audio
action manipulate
sdp-line-rule
name check_payload
type m
action reject
comparison-type pattern-rule
match-value ^(audio) ( [0-9]{4,5}) ( UDP 0)$
new-value "403:Not Acceptable Media"

sip-manipulation
name ToCUCM
header-rule
name ForNAT_IP
header-name From
action sip-manip
new-value Topohiding
header-rule
name forRURI
header-name From
action sip-manip
new-value ModRURItocUCM
header-rule
name ForupdatetocUCM
header-name From
action sip-manip
new-value ModSupportedtowardsCUCM

sip-manipulation
name Topohiding
header-rule
name From
header-name From
action manipulate
element-rule
name From_header
type uri-host
action replace
new-value $LOCAL_IP
header-rule
name To
header-name To
action manipulate
element-rule
name To
type uri-host
action replace

```

```

new-value                                $REMOTE_IP

sip-manipulation
  name                                    anonymouscall
  header-rule
    name                                    changeRURI
    header-name                            Request-URI
    action                                  manipulate
    msg-type                                request
    methods                                  INVITE
    element-rule
      name                                    storeuser
      type                                    uri-user
      action                                  store
      comparison-type                        pattern-rule
      match-value                            ^\+184 (.*$)
    element-rule
      name                                    striptheplus
      type                                    uri-user
      action                                  replace
      comparison-type                        boolean
      match-value                            $changeRURI.$storeuser
      new-value                              $ORIGINAL-^"+"

  header-rule
    name                                    addphonecontext
    header-name                            Request-URI
    action                                  manipulate
    comparison-type                        boolean
    msg-type                                request
    methods                                  INVITE
    match-value                            $changeRURI.$storeuser.$0
    element-rule
      name                                    addtheparam
      parameter-name                        phone-context
      type                                    uri-user-param
      action                                  add
      new-value                              \+81

  header-rule
    name                                    ModToheader
    header-name                            To
    action                                  manipulate
    comparison-type                        pattern-rule
    msg-type                                request
    methods                                  INVITE
    element-rule
      name                                    storetheuser
      type                                    uri-user
      action                                  store
      comparison-type                        pattern-rule
      match-value                            ^\+184 (.*$)
    element-rule
      name                                    Striptheplusfromuriuser
      type                                    uri-user
      action                                  replace
      comparison-type                        boolean
      match-value                            $ModToheader.$storetheuser
      new-value                              $ORIGINAL-^"+"

  header-rule
    name                                    addphonecontextinTo
    header-name                            To
    action                                  manipulate

```



```

        comparison-type      boolean
        msg-type             request
        methods              INVITE
        match-value          $ModToheader.$storetheuser.$0
        element-rule
            name              addpc
            parameter-name    phone-context
            type              uri-user-param
            action            add
            new-value         \+81

sip-manipulation
    name                    changeforPCPID
    header-rule
        name                modforPCPID
        header-name         To
        action              manipulate
        comparison-type     pattern-rule
        msg-type            out-of-dialog
        methods              INVITE
        element-rule
            name              modToer
            type              uri-user
            action            replace
            comparison-type  pattern-rule
            match-value      !($TO_USER.$0 == $PCPID_USER.$0)
            new-value        $PCPID_USER.$0

sip-manipulation
    name                    checkPCPID
    header-rule
        name                StorePCPIDuser
        header-name         P-Called-Party-ID
        action              manipulate
        comparison-type     pattern-rule
        element-rule
            name              Storeuser
            type              header-value
            action            sip-manip
            comparison-type  pattern-rule
            new-value        changeforPCPID

sip-manipulation
    name                    checkTotag
    header-rule
        name                storeTotag
        header-name         To
        action              store
        comparison-type     pattern-rule
        match-value         sip:*
        element-rule
            name              Storetag
            parameter-name    tag
            type              header-param
            action            store

    header-rule
        name                checkTotag
        header-name         To
        action              manipulate
        comparison-type     boolean
        match-value         $storeTotag.$Storetag
        element-rule
            name              reinv

```

	parameter-name	tag
	type	header-param
	action	sip-manip
	new-value	rejectreinv
sip-manipulation		
name	checkforBYE	
header-rule		
name	storeBYE	
header-name	To	
action	manipulate	
comparison-type	pattern-rule	
methods	BYE	
element-rule		
name	storetheuser	
type	uri-user	
action	store	
comparison-type	pattern-rule	
header-rule		
name	checkBYE	
header-name	Request-URI	
action	manipulate	
comparison-type	pattern-rule	
methods	BYE	
element-rule		
name	checkBYERURI	
type	uri-user	
action	reject	
comparison-type	boolean	
match-value	!\$storeBYE.\$storetheuser	
new-value	404:Not Found	
sip-manipulation		
name	convertresp	
header-rule		
name	change503to580	
header-name	@status-line	
action	manipulate	
msg-type	reply	
element-rule		
name	modStatusCode	
type	status-code	
action	replace	
comparison-type	pattern-rule	
match-value	(503 403)	
new-value	580	
sip-manipulation		
name	deltransportUDP	
header-rule		
name	Remtransportudp	
header-name	Contact	
action	manipulate	
methods	INVITE,UPDATE	
element-rule		
name	delparam	
parameter-name	transport	
type	uri-param	
action	delete-element	
sip-manipulation		
name	drop100	
header-rule		
name	checkRURI	

```

header-name Request-URI
action manipulate
msg-type reply
methods INVITE
element-rule
    name storemsgcode
    type status-code
    action store
    comparison-type pattern-rule
    match-value 100
element-rule
    name dropthemessage
    type header-value
    action reject
    match-value $checkRURI.$storemsgcode.$0
sip-manipulation
    name dropACK
    header-rule
        name rejectACK
        header-name To
        action reject
        comparison-type pattern-rule
        msg-type request
        methods ACK
        match-value !($RURI_USER.$0 == $TO_USER.$0)
        new-value "698:Match Not Found"
sip-manipulation
    name forsessionexpirestoNTT
    header-rule
        name adduacforSE
        header-name Session-Expires
        action manipulate
        comparison-type pattern-rule
        msg-type request
        methods INVITE
        element-rule
            name storesevalue
            type header-value
            action store
            comparison-type pattern-rule
            match-value (.*)
        element-rule
            name addrefresheruac
            type header-value
            action replace
            comparison-type pattern-rule
            new-value 180+;+refresher=uac
sip-manipulation
    name inboundNTT
    header-rule
        name changeToperPCPID
        header-name From
        action sip-manip
        methods INVITE
        new-value checkPCPID
    header-rule
        name TC1282hr
        header-name From
        action sip-manip
        methods INVITE

```

```

    new-value TC1282
header-rule
  name TC1283hr
  header-name From
  action sip-manip
  msg-type request
  new-value TC1283
header-rule
  name TC1284hr
  header-name From
  action sip-manip
  methods INVITE
  new-value TC1284
header-rule
  name rejPRACK
  header-name From
  action sip-manip
  methods BYE,PRACK
  new-value rejectPRACK
header-rule
  name tc2215reinv
  header-name From
  action sip-manip
  msg-type request
  new-value checkTotag
header-rule
  name stoprecurseforreg
  header-name From
  action sip-manip
  msg-type reply
  new-value regrecurse
header-rule
  name hmrforack
  header-name From
  action sip-manip
  msg-type request
  methods ACK
  new-value dropACK
sip-manipulation
  name prackwork
  header-rule
    name delsupported
    header-name Supported
    action delete
    msg-type request
    methods INVITE
  header-rule
    name addrequireinINVITE
    header-name Require
    action add
    msg-type request
    methods INVITE
    new-value 100rel
sip-manipulation
  name regrecurse
  header-rule
    name checkcseq
    header-name CSeq
    action sip-manip
    comparison-type pattern-rule

```

```

        msg-type                reply
        match-value              ([0-9]{1} REGISTER)
        new-value                 convertresp
sip-manipulation
  name                          rejectPRACK
  header-rule
    name                        rejectPRACK
    header-name                  To
    action                       reject
    comparison-type              pattern-rule
    msg-type                     request
    methods                      BYE,PRACK
    match-value                  !($RURI_USER.$0 == $TO_USER.$0)
    new-value                     "481:Call/Transaction Does Not Exist"
sip-manipulation
  name                          rejectreinv
  header-rule
    name                        rejectreinv
    header-name                  To
    action                       reject
    comparison-type              pattern-rule
    msg-type                     request
    methods                      INVITE
    match-value                  !($RURI_USER.$0 == $TO_USER.$0)
    new-value                     "404:Not found"
sip-manipulation
  name                          removecallinfo
  header-rule
    name                        Stripcallinfo
    header-name                  Call-Info
    action                       delete
    methods                      INVITE,UPDATE
sip-monitoring
  match-any-filter              enabled
spl-config
  spl-options                   log-sip-msg
  plugins
    name                        ocSurrogateRegister-1-6.pkg
  plugins
    name                        ocNttMsgConverter-0-3.pkg
steering-pool
  ip-address                    10.232.50.60
  start-port                    35000
  end-port                      37000
  realm-id                      CUCM
steering-pool
  ip-address                    172.16.1.21
  start-port                    41000
  end-port                      45000
  realm-id                      NTT-router
surrogate-agent
  register-host                 ntt.com
  register-user                 +81XXXXXXXXXX
  realm-id                      CUCM
  customer-next-hop             ntt.com
  register-contact-host         172.16.1.21
  register-contact-user         +81XXXXXXXXXX
  password                      *****
  register-expires              3600

```

```
    auth-user                user
    register-retry-time      1800
system-config
    process-log-level        DEBUG
    comm-monitor
        state                enabled
        sbc-grp-id           5
    default-gateway          172.16.1.15
translation-rules
    id                       addforJP
    type                     add
    add-string               +
translation-rules
    id                       remove81
    type                     delete
    delete-string           +
web-server-config
```

Troubleshooting Tools

If you find that you are not able to complete calls or have problems with the test cases, there are a few tools available for Windows Server, Lync Server, and the Oracle SBC like logging and tracing which may be of assistance. In this section we will provide a list of tools which you can use to aid in troubleshooting any issues you may encounter.

Since we are concerned with communication between the Lync Server mediation server and the SBC we will focus on the troubleshooting tools to use between those devices if calls are not working or tests are not passing.

Microsoft Network Monitor (NetMon)

NetMon is a network protocol analyzer which is freely downloadable from Microsoft. It can be found at www.microsoft.com/downloads. NetMon could be installed on the Lync Server mediation server, the Lync Server Standard Edition server, or Enterprise Edition front end server.

Wireshark

Wireshark is also a network protocol analyzer which is freely downloadable from www.wireshark.org. Wireshark could be installed on the Lync Server mediation server, the Lync Server Standard Edition server, or MCS Enterprise Edition front end server.

Eventviewer

There are several locations in the event viewer where you can find valuable information to aid in troubleshooting issues with your deployment.

With the requirement that there is a completely functioning Lync Server with Enterprise Voice deployment in place, there are only a few areas in which one would use the Event Viewer for troubleshooting:

- The Enterprise Voice client;
- The Lync Server Front End server;
- A Lync Server Standard Edition Server; and
- A Lync Server Mediation Server.

On the Oracle SBC 4600 Series

The Oracle SBC provides a rich set of statistical counters available from the ACLI, as well as log file output with configurable detail. The follow sections detail enabling, adjusting and accessing those interfaces.

Resetting the statistical counters, enabling logging and restarting the log files.

At the SBC Console:

```
oraclesbcl# reset sipd
oraclesbcl# notify sipd debug
oraclesbcl#
enabled SIP Debugging
oraclesbcl# notify all rotate-logs
```

Examining the log files

Note: You will FTP to the management interface of the SBC with the username user and user mode password (the default is “acme”).

```
C:\Documents and Settings\user>ftp 192.168.5.24
Connected to 192.168.85.55.
220 oraclesbc1FTP server (VxWorks 6.4) ready.
User (192.168.85.55:(none)): user
331 Password required for user.
Password: acme
230 User user logged in.
ftp> cd /ramdrv/logs
250 CWD command successful.
ftp> get sipmsg.log
200 PORT command successful.
150 Opening ASCII mode data connection for '/ramdrv/logs/sipmsg.log' (3353
bytes).
226 Transfer complete.
ftp: 3447 bytes received in 0.00Seconds 3447000.00Kbytes/sec.
ftp> get log.sipd
200 PORT command successful.
150 Opening ASCII mode data connection for '/ramdrv/logs/log.sipd' (204681
bytes).
226 Transfer complete.
ftp: 206823 bytes received in 0.11Seconds 1897.46Kbytes/sec.
ftp> bye
221 Goodbye.
```

You may now examine the log files with the text editor of your choice.

Through the Web GUI

You can also check the display results of filtered SIP session data from the Oracle Enterprise Session Border Controller, and provides traces in a common log format for local viewing or for exporting to your PC. Please check the “Monitor and Trace” section (page 145) of the Web GUI User Guide available at http://docs.oracle.com/cd/E56581_01/index.htm

Telnet

Since we are working within an architecture which uses bound TCP listening ports for functionality, the simplest form of troubleshooting can be seeing if the devices are listening on a particular port, as well as confirming that there is nothing blocking them such as firewalls. Ensure that you have a TELNET client available on a workstation as well as on the Lync Server mediation server.

The Lync Server mediation server will listen on TCP port 5067 by default for SIP signaling. In our example we are listening on 5060 on the PSTN facing NIC. From the Standard Edition pool or Enterprise Edition pool the Mediation Server would be listening on port 5061. Tests may include:

- Client to pool server: **telnet <servername> 5061**
- Pool server to Mediation Server: **telnet <servername> 5061**

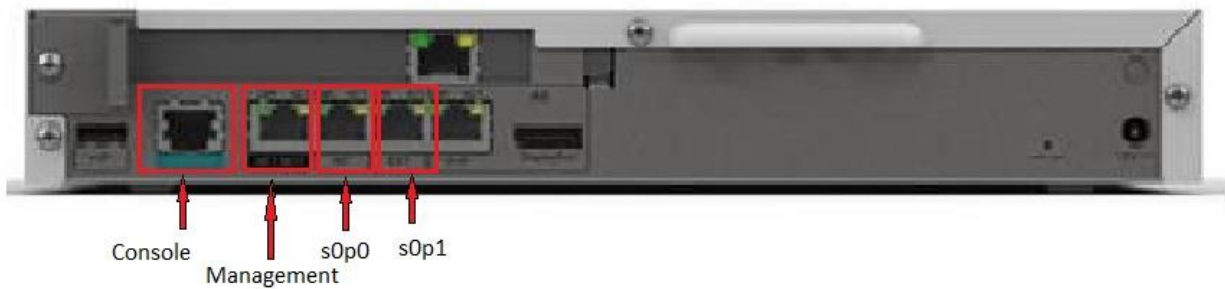
Appendix A

Accessing the ACLI

Access to the ACLI is provided by:

- The serial console connection;
- TELNET, which is enabled by default but may be disabled; and
- SSH, this must be explicitly configured.

Initial connectivity will be through the serial console port. At a minimum, this is how to configure the management (eth0) interface on the SBC.

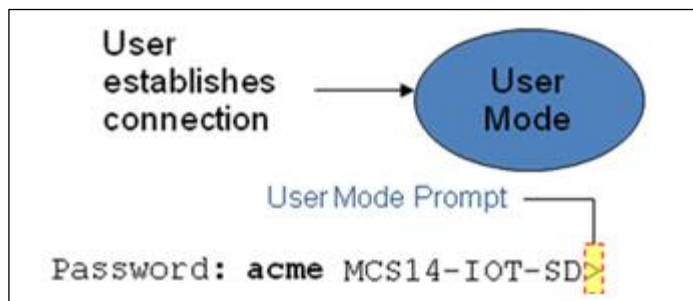


ACLI Basics

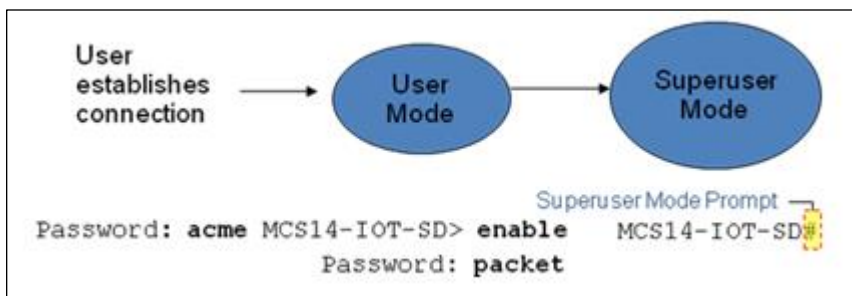
There are two password protected modes of operation within the ACLI, User mode and Superuser mode.

When you establish a connection to the SBC, the prompt for the User mode password appears. The default password is acme.

User mode consists of a restricted set of basic monitoring commands and is identified by the greater than sign (>) in the system prompt after the target name. You cannot perform configuration and maintenance from this mode.



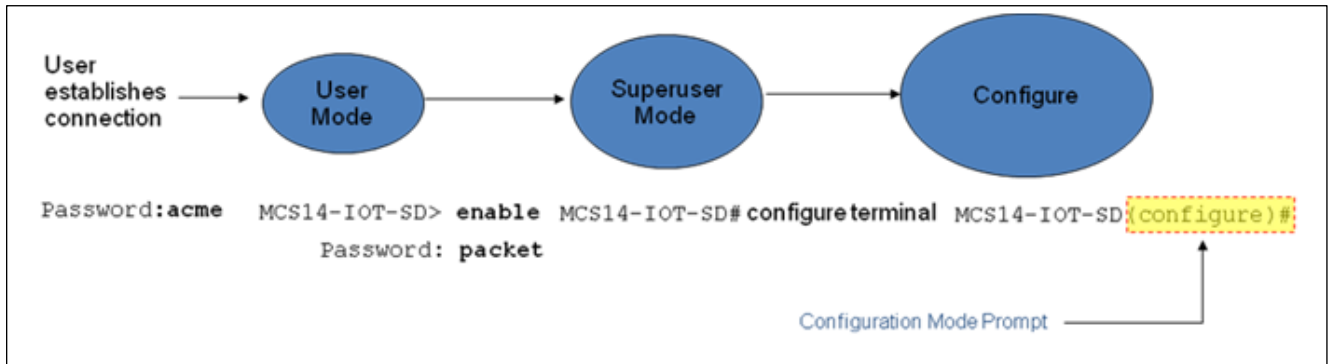
The Superuser mode allows for access to all system commands for operation, maintenance, and administration. This mode is identified by the pound sign (#) in the prompt after the target name. To enter the Superuser mode, issue the enable command in the User mode.



From the Superuser mode, you can perform monitoring and administrative tasks; however you cannot configure any elements. To return to User mode, issue the exit command.

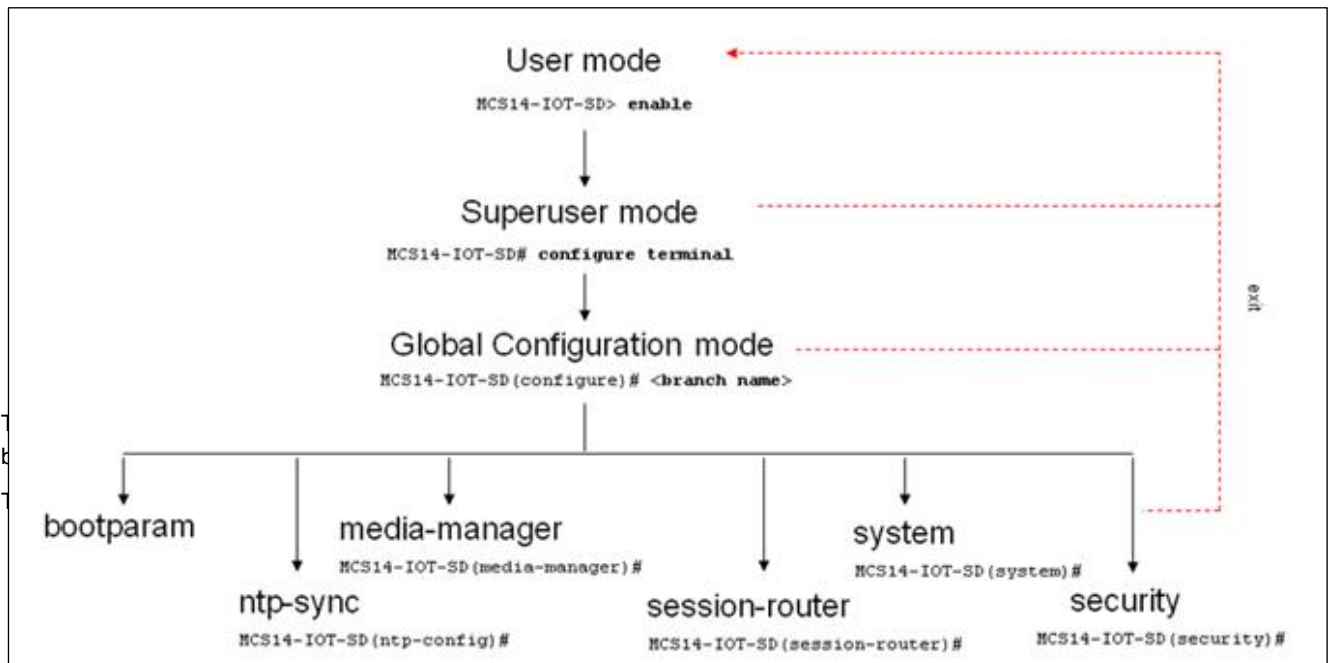
You must enter the Configuration mode to configure elements. For example, you can access the configuration branches and configuration elements for signaling and media configurations. To enter the Configuration mode, issue the **configure terminal** command in the Superuser mode.

Configuration mode is identified by the word configure in parenthesis followed by the pound sign (#) in the prompt after the target name, for example, **oraclesbc1(configure)#**. To return to the Superuser mode, issue the **exit** command.



In the configuration mode, there are six configuration branches:

- bootparam;
- ntp-sync;
- media-manager;
- session-router;
- system; and
- security.



- host inet –The IP address of external server where image file resides.
- user and ftp password – Used to boot from the external FTP server.
- gateway inet – The gateway IP address for reaching the external server, if the server is located in a different network.

```

'.' = clear field;  '-' = go to previous field;  q = quit
boot device          : eth0
processor number     : 0
host name            :
file name            : /tffs0/nnSCX620.gz
inet on ethernet (e) : 10.0.3.11:ffff0000
inet on backplane (b) :
host inet (h)        : 10.0.3.100
gateway inet (g)     : 10.0.0.1
user (u)             : anonymous
ftp password (pw) (blank = rsh) : anonymous
flags (f)            : 0x8
target name (tn)     : MCS14-IOT-SD
startup script (s)   :
other (o)

```

The ntp-sync branch provides access to ntp server configuration commands for synchronizing the SBC time and date.

The security branch provides access to security configuration.

The system branch provides access to basic configuration elements as system-config, snmp-community, redundancy, physical interfaces, network interfaces, etc.

The session-router branch provides access to signaling and routing related elements, including H323-config, sip-config, iwf-config, local-policy, sip-manipulation, session-agent, etc.

The media-manager branch provides access to media-related elements, including realms, steering pools, dns-config, media-manager, and so forth.

You will use media-manager, session-router, and system branches for most of your working configuration.

Configuration Elements

The configuration branches contain the configuration elements. Each configurable object is referred to as an element. Each element consists of a number of configurable parameters.

Some elements are single-instance elements, meaning that there is only one of that type of the element - for example, the global system configuration and redundancy configuration.

Some elements are multiple-instance elements. There may be one or more of the elements of any given type. For example, physical and network interfaces.

Some elements (both single and multiple instance) have sub-elements. For example:

- SIP-ports - are children of the sip-interface element
- peers – are children of the redundancy element
- destinations – are children of the peer element

Creating an Element

1. To create a single-instance element, you go to the appropriate level in the ACLI path and enter its parameters. There is no need to specify a unique identifier property because a single-instance element is a global element and there is only one instance of this element.

2. When creating a multiple-instance element, you must specify a unique identifier for each instance of the element.
3. It is important to check the parameters of the element you are configuring before committing the changes. You do this by issuing the `show` command before issuing the `done` command. The parameters that you did not configure are filled with either default values or left empty.
4. On completion, you must issue the `done` command. The `done` command causes the configuration to be echoed to the screen and commits the changes to the volatile memory. It is a good idea to review this output to ensure that your configurations are correct.
5. Issue the `exit` command to exit the selected element.

Note that the configurations at this point are not permanently saved yet. If the SBC reboots, your configurations will be lost.

Editing an Element

The procedure of editing an element is similar to creating an element, except that you must select the element that you will edit before editing it.

1. Enter the element that you will edit at the correct level of the ACLI path.
2. Select the element that you will edit, and view it before editing it.
The `select` command loads the element to the volatile memory for editing. The `show` command allows you to view the element to ensure that it is the right one that you want to edit.
3. Once you are sure that the element you selected is the right one for editing, edit the parameter one by one. The new value you provide will overwrite the old value.
4. It is important to check the properties of the element you are configuring before committing it to the volatile memory. You do this by issuing the `show` command before issuing the `done` command.
5. On completion, you must issue the `done` command.
6. Issue the `exit` command to exit the selected element.

Note that the configurations at this point are not permanently saved yet. If the SBC reboots, your configurations will be lost.

Deleting an Element

The `no` command deletes an element from the configuration in editing.

To delete a single-instance element,

1. Enter the `no` command from within the path for that specific element
2. Issue the `exit` command.

To delete a multiple-instance element,

1. Enter the `no` command from within the path for that particular element.
The key field prompt, such as `<name>:<sub-port-id>`, appears.
2. Use the `<Enter>` key to display a list of the existing configured elements.
3. Enter the number corresponding to the element you wish to delete.
4. Issue the `select` command to view the list of elements to confirm that the element was removed.

Note that the configuration changes at this point are not permanently saved yet. If the SBC reboots, your configurations will be lost.

Configuration Versions

At any time, three versions of the configuration can exist on the SBC: the edited configuration, the saved configuration, and the running configuration.

- The **edited configuration** – this is the version that you are making changes to. This version of the configuration is stored in the SBC’s volatile memory and will be lost on a reboot.
To view the editing configuration, issue the `show configuration` command.
- The **saved configuration** – on issuing the `save-config` command, the edited configuration is copied into the non-volatile memory on the SBC and becomes the saved configuration. Because the saved configuration has not been activated yet, the changes in the configuration will not take effect. On reboot, the last activated configuration (i.e., the last running configuration) will be loaded, not the saved configuration.
- The **running configuration** is the saved then activated configuration. On issuing the `activate-config` command, the saved configuration is copied from the non-volatile memory to the volatile memory. The saved configuration is activated and becomes the running configuration. Although most of the configurations can take effect once being activated without reboot, some configurations require a reboot for the changes to take effect.
To view the running configuration, issue command `show running-config`.

Saving the Configuration

The `save-config` command stores the edited configuration persistently.

Because the saved configuration has not been activated yet, changes in configuration will not take effect. On reboot, the last activated configuration (i.e., the last running configuration) will be loaded. At this stage, the saved configuration is different from the running configuration.

Because the saved configuration is stored in non-volatile memory, it can be accessed and activated at later time.

Upon issuing the `save-config` command, the SBC displays a reminder on screen stating that you must use the `activate-config` command if you want the configurations to be updated.

```

oraclesbcl # save-config
Save-Config received, processing.
waiting 1200 for request to finish
Request to 'SAVE-CONFIG' has Finished,
Save complete
Currently active and saved configurations do not match!
To sync & activate, run 'activate-config' or 'reboot activate'.
oraclesbcl #

```

Activating the Configuration





On issuing the `activate-config` command, the saved configuration is copied from the non-volatile memory to the volatile memory. The saved configuration is activated and becomes the running configuration.

Some configuration changes are service affecting when activated. For these configurations, the SBC warns that the change could have an impact on service with the configuration elements that will potentially be service affecting. You may decide whether or not to continue with applying these changes immediately or to apply them at a later time.

```
oraclesbcl# activate-config
Activate-Config received, processing.
waiting 120000 for request to finish
Request to 'ACTIVATE-CONFIG' has Finished,
Activate Complete
oraclesbcl#
```



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Integrated Cloud Applications & Platform Services

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