



Oracle Enterprise Communications Broker &
Oracle Enterprise Session Border Controller
with Avaya Aura 7.0, Cisco's UCM 11.0,
Microsoft's Lync2013, Skype for Business &
CCE

Technical Application Note



Disclaimer

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Table of Contents

INTENDED AUDIENCE	5
DOCUMENT OVERVIEW	5
INTRODUCTION	6
REQUIREMENTS.....	6
COMPONENTS, VERSIONS AND DEBUGGING TOOLS	6
LAB CONFIGURATION	7
CONFIGURING THE ORACLE ENTERPRISE SBC	8
IN SCOPE.....	8
OUT OF SCOPE	8
WHAT WILL YOU NEED.....	8
SBC- GETTING STARTED	8
Establish the serial connection and logging in the SBC.....	9
Initial Configuration – Assigning the management Interface an IP address	9
CONFIGURING THE SBC	10
High Availability	10
Local Policies.....	10
Media Manager.....	13
Network Interfaces.....	13
Physical Interfaces.....	14
Realm Configs	14
Playback Config.....	16
Session Agents.....	17
Session Translation	18
SIP Config.....	18
SIP Interfaces.....	18
SIP Manipulations (Header Manipulation Rules – HMR)	20
SIP Monitoring.....	28
Steering Pools.....	28
Certificate record	29
TLS profile.....	30
Tls Global config	30
Sdes Profile	31
Media-sec-policy.....	31
RTCP Policy.....	31
Response Map.....	31
LDAP Config.....	32
Codec-policy.....	32
System Config	33
Translation Rules	33
Web Server Config	33
Save, Activate, and Reboot	33
LYNC 2013(10.70.50.20)	35
Topology builder with PSTN gateway.....	35
Lync 2013 Control panel with “Associated PSTN gateway”	37
AVAYA (10.70.50.21)	39

SBC Configured on Session Manger.....	39
Routing Policy.....	40
Dial Pattern.....	41
CISCO UCM (10.70.50.22).....	42
SIP Trunk List.....	42
SIP Trunk Configuration.....	43
Route Pattern.....	46
Route Pattern for 10 Digit Dialing.....	47
Route Pattern for Extension Dialing.....	48
CLOUD CONNECTER EDITION (10.64.4.149).....	49
TEST PLAN & RESULTS.....	50
TEST PLAN.....	50
APPENDIX A.....	55
FULL SBC CONFIGURATION.....	55
TROUBLESHOOTING TOOLS.....	72
MICROSOFT NETWORK MONITOR (NETMON).....	72
WIRESHARK.....	72
EVENTVIEWER.....	72
ON THE ORACLE ECB AND E-SBC.....	72
Resetting the statistical counters, enabling logging and restarting the log files.....	72
Examining the log files.....	72
Through the Web GUI.....	73
TELNET.....	73
CISCO REAL-TIME MONITORING TOOL (RTMT).....	73
APPENDIX B.....	74
ACCESSING THE ACLI.....	74
ACLI BASICS.....	74
CONFIGURATION ELEMENTS.....	76
CREATING AN ELEMENT.....	76
EDITING AN ELEMENT.....	76
DELETING AN ELEMENT.....	77
CONFIGURATION VERSIONS.....	77
SAVING THE CONFIGURATION.....	77
ACTIVATING THE CONFIGURATION.....	78



Intended Audience

This is a technical document intended for telecommunications engineers with the purpose of configuring the Oracle Communications Enterprise-SBC, Microsoft Skype for Business, Avaya Aura Session Manager and Cisco Unified Communications Manager. There will be steps that require navigating Microsoft Windows Server 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.

Document Overview

This technical application note documents the implementation of the Oracle Enterprise Session Border Controller (ESBC) in an Enterprise network consisting of multi-vendor Unified Communications platforms – Microsoft's Cloud Connector Edition & Skype for Business 2015, Avaya Aura Session Manager and Cisco Unified Communications Manager - connecting to a SIP trunk.

Introduction

Requirements

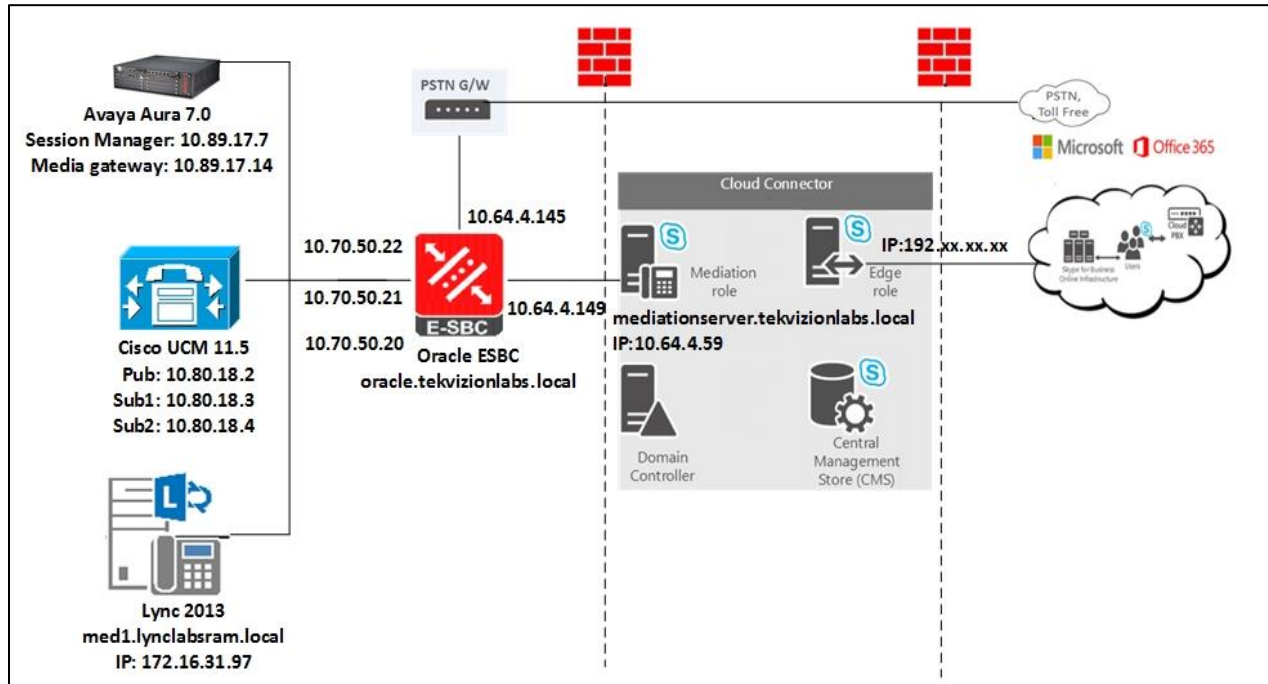
- Oracle Enterprise Session Border Controller ECZ750 or higher
- Microsoft Skype for Business 2015
- Microsoft Lync 2013
- Microsoft Cloud Connector Edition
- Avaya Aura 7.0
- Cisco Unified Communications Manager 11.0
- SIP Trunk (Verizon)

Components, Versions and Debugging Tools

Components and Debugging Tools	Version
E-SBC	Acme Packet 3900 ECZ7.5.0 or above
Microsoft Lync 2013 Server	5.0.8308.872
Microsoft Lync 2013 Client	15.0.4551.1007
Microsoft Skype for Business Cloud Server	6.0.9319.377
Microsoft Cloud Connector Edition	1.4.2
Microsoft Skype for Business 2015 Client	16.0.4417.1000
Cisco Unified Communication Manager	11.5.1.12900-21
Avaya Aura	7.0
Wireshark	2.2.7
CLS Logger for Skype for Business	6.0.9319.0

Lab Configuration

The following diagram illustrates the lab environment created by tekVizion to facilitate certification testing. tekVizion is a Certification house dedicated to the telecommunications industry. Their core services include consulting/solution design, interoperability/verification testing, integration, custom software development and solution support services.



The network architecture consists of two areas. Area 1 represents the Enterprise network and Area 2 is the service provide network. The Enterprise network has the ESBC at its core connecting together multiple UC platforms.

Configuring the Oracle Enterprise SBC

In this section we describe the steps for configuring an Oracle Enterprise SBC (E-SBC) for use with the Oracle ECB, Microsoft CCE & Skype for Business, Cisco CUCM, and Avaya Aura. The E-SBC will connect the Enterprise network to the Service Provider network in a SIP trunking scenario.

In Scope

The following guide for configuring the Oracle SBC assumes that this is a newly deployed device dedicated to a single customer. 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 the SBC. This document covers the setup for the 1100, 3820, 4500, 4600, and 6300 platforms running OS ECZ7.3.0 MR-1 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
- Configuration of Distributed Denial of Service (DDoS) protection parameters as these are based on individual customer requirements.

What will you need

- RJ45/DB9 serial adapter provided with the SBC, along with a straight-through Ethernet cable to go from the adapter to the SBC's console port (on the rear of the 1100, 4600, and 6300, and the front of the 3820 and 4500).
- 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 the management interface (eth0, labeled Mgmt0 on the SBC chassis) of the SBC - the eth0 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 compromised DDoS protection. Oracle does not support SBC configurations with management and media/service interfaces on the same subnet.
- IP address of the Oracle ECB.
- IP addresses to be used for the SBC internal and external facing ports (Service Interfaces)

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, on platforms such as the 3820 and 4500 that have two console ports.**

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

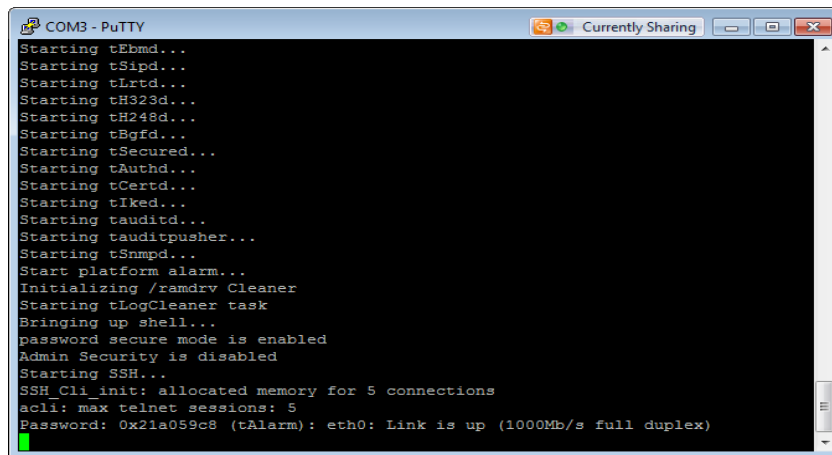
All commands are in bold, such as **configure terminal**; parameters in bold red such as **oraclesbc1** are parameters which are specific to an individual deployment. **Note:** The CLI is case sensitive.

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 console port on the SBC and the other end to console adapter that ships with the SBC, connect the console adapter (a DB9 adapter) to the DB9 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 tEbmd...
Starting tSipd...
Starting tLtd...
Starting tH323d...
Starting tH248d...
Starting tBgfd...
Starting tSecured...
Starting tAuthd...
Starting tCerd...
Starting tKed...
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
acl: 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. **Note these example boot parameters are specific to the 4600 platform. Other platforms will have different boot parameters. Use nnECZ730m1.64.bz for the 1100, 4500, 4600, and the 6300. Use nnECZ730m1.32.bz for the 3820.**

```
ORACLESBC1(configure)# bootparam

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

```

Boot File           : /boot/nxECZ750.bz
IP Address          : 192.168.79.44
VLAN                :
Netmask             : 255.255.255.224
Gateway            : 192.168.79.33
IPv6 Address        :
IPv6 Gateway        :
Host IP             : 0.0.0.0
FTP username        : vxftp
FTP password        : vxftp123
Flags               :
Target Name         : oraclesbc1
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 Enterprise SBC in an environment with Microsoft Lync, Skype for Business, Cisco CUCM, and Avaya Aura.

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

Please note – All the configs sections contain output of “show running-config short” – the short command only prints out non-standard configs required for the SBC to operate in the model configured

High Availability

The Mgmt1 and Mgmt2 (labeled wancom1 and wancom2 in the configuration) ports which are on the rear panel of the SBC are used for the purpose of High Availability on the E-SBC. Crossover cables must be connected between these ports on the SBCs, i.e. Mgmt1 to Mgmt1 and Mgmt2 to Mgmt2. Please refer to the “High Availability Nodes” in the ACLI configuration guide for ECZ730 for more details.

Local Policies

Path: `configure terminal > session-router > local-policy`

Policy number	Trigger	Destination	Logic
1	25/52600025	Avaya	<ul style="list-style-type: none"> All extensions which start with 25 – would terminate on Avaya cluster If an external (outside of the local network) user dials 52600025 – those calls also would land on Avaya Cluster
2	26/52600026	CUCM	<ul style="list-style-type: none"> All extensions which start with 26 – would terminate on CUCM cluster

			<ul style="list-style-type: none"> If an external (outside of the local network) user dials 52600026 – those calls also would land on CUCM Cluster
3	27/52600027	Lync 2013	<ul style="list-style-type: none"> All extensions which start with 27 – would terminate on SFB medPool If an external (outside of the local network) user dials 52600027 – those calls also would land on Mediation server pool
4	28/52600028	O365	<ul style="list-style-type: none"> All extensions which start with 28 – would terminate on SFB online running on Office 365 – these calls would route via CCE If an external (outside of the local network) user dials 52600028 – those calls also would land on Office365 medPool
5	*	PSTN	<ul style="list-style-type: none"> Any calls sourced by Avaya/CUCM/SFB/CCE realms would be delivered to PSTN
6	2432 2601 5270002601	LDAP	<ul style="list-style-type: none"> The following numbers 2432/2601/5270002601 were configured specifically for LDAP based routing – 2432 was a Avaya user, 2601 & 5270002601 were CUCM users.
7	medl.lyncclabsram.local	Lync2013	<ul style="list-style-type: none"> This policy was created to assist with refer-call-transfer feature where the SBC terminates REFER message and sets up a new call based on refer-to header.
8	mediationserver.tekvizionlabs.local	CCE	<ul style="list-style-type: none"> This policy was created to assist with refer-call-transfer feature where the SBC terminates REFER message and sets up a new call based on refer-to header
9	9876543210	SFB	<ul style="list-style-type: none"> All calls made to 9876543210 would be routed to SFB
10	*	Lync2013	<ul style="list-style-type: none"> All OPTIONS messages that landed on the SBC would be routed to Lync 2013 for proper OPTIONS handling.

Please Note – the numbers in brackets below are only for reference for the table above

(1) local-policy	
from-address	*
to-address	25 52600025
source-realm	*
policy-attribute	
next-hop	10.89.17.7
realm	AvayaRealm
(2) local-policy	
from-address	*
to-address	26 52700026
source-realm	*
policy-attribute	

	next-hop	10.80.18.3
	realm	CUCMRealm
(3)	local-policy	
	from-address	*
	to-address	27
		52800027
	source-realm	*
	policy-attribute	
	next-hop	med1.lyncclabsram.local
	realm	inside
(4)	local-policy	
	from-address	*
	to-address	28
		52900028
	source-realm	*
	policy-attribute	
	next-hop	mediationserver.tekvizionlabs.local
	realm	outside
(5)	local-policy	
	from-address	*
	to-address	2142425
		800
		9725
	source-realm	*
	description	topstn
	policy-attribute	
	next-hop	10.64.1.72
	realm	PSTN
(6)	local-policy	
	from-address	*
	to-address	2432
		2601
		5270002601
	source-realm	*
	policy-attribute	
	next-hop	ldap:ldapCfg
	realm	inside
(7)	local-policy	
	from-address	*
	to-address	med1.lyncclabsram.local
	source-realm	*
	policy-attribute	
	next-hop	med1.lyncclabsram.local
	realm	inside
(8)	local-policy	
	from-address	*
	to-address	mediationserver.tekvizionlabs.local
	source-realm	*
	policy-attribute	
	next-hop	mediationserver.tekvizionlabs.local
	realm	outside
(9)	local-policy	
	from-address	*
	to-address	9876543210
	source-realm	*
	policy-attribute	

```

        next-hop          172.16.29.71
        realm             toSFB
(10) local-policy
    from-address         *
    to-address           *
    source-realm         outside
    policy-attribute
        next-hop          med1.lyncclabsram.local
        realm             inside
        methods           OPTIONS

```

Media Manager

Path: **configure terminal > media-manager > media-manager > select > done**

```

media-manager
    initial-guard-timer      86400
    options                  xcode-gratuitous-rtcp-report-generation

```

Network Interfaces

Path: **configure terminal > system > network-interface**

```

network-interface
    name                    s0p0
    description              WAN
    ip-address               10.64.4.149
    netmask                  255.255.0.0
    gateway                  10.64.1.1
    gw-heartbeat
        state                enabled
    dns-ip-primary           10.64.4.61
    dns-domain                tekvizionlabs.local
    hip-ip-list              10.64.4.149
                            10.64.4.145
                            10.64.4.147
    icmp-address             10.64.4.149
                            10.64.4.145
                            10.64.4.147

network-interface
    name                    s0p1
    description              LAN
    ip-address               10.70.50.20
    netmask                  255.255.255.0
    gateway                  10.70.50.1
    dns-ip-primary           172.16.31.91
    dns-domain                lyncclabsram.local
    hip-ip-list              10.70.50.20
                            10.70.50.21
                            10.70.50.22
                            10.70.50.23
    icmp-address             10.70.50.20
                            10.70.50.21
                            10.70.50.22
                            10.70.50.23

```

Physical Interfaces

Path: **configure terminal > system > phy-interface**

```

phy-interface
  name                s0p0
  operation-type      Media
phy-interface
  name                s0p1
  operation-type      Media
  port                1
  
```

Realm Configs

Path: **configure terminal > media-manager > realm-config**

Realm	Audio	Transcoding	RTCP generation
Avaya	RTP	CN generation enabled	Yes
CUCM	RTP	G729 only, no PCMU/PCMA	Yes
PSTN	RTP	G711 only	Yes
Inside (Lync2013)	RTP	G711 only	Yes
Outside (Office365 SFB)	SRTP	G711 only, CN generation enabled	Yes
Outside-rtp (Office365 SFB)	RTP	G711 only	No
toSFB	SRTP	G711 only, CN generation enabled	Yes

```

realm-config
  identifier           AvayaRealm
  network-interfaces  s0p1:0
  media-sec-policy    RTP
  early-media-allow   none
  codec-policy        Xcode
  rtcp-policy         rtcpGen
realm-config
  identifier           CUCMRealm
  network-interfaces  s0p1:0
  media-sec-policy    RTP
  codec-policy        G729only
  codec-manip-in-realm enabled
  rtcp-policy         rtcpGen
realm-config
  identifier           PSTN
  network-interfaces  s0p0:0
  media-sec-policy    RTP
  codec-policy        G711Only
  rtcp-policy         rtcpGen
realm-config
  
```

identifier	inside
network-interfaces	s0p1:0
media-sec-policy	RTP
spl-options	comfort-noise-generate
codec-policy	G711Only
rtcp-policy	rtcpGen
realm-config	
identifier	outside
network-interfaces	s0p0:0
media-sec-policy	SRTP
spl-options	comfort-noise-generate
codec-policy	G711Only
rtcp-policy	rtcpGen
realm-config	
identifier	outside-rtp
network-interfaces	s0p0:0
realm-config	
identifier	toSFB
network-interfaces	s0p1:0
media-sec-policy	RTP
codec-policy	Xcode
rtcp-policy	rtcpGen

Playback Config

Path: `configure terminal > media-manager > playback-config`

We utilize the local playback feature of the SBC to play ring back tone during transfers. The ringback tone is played based on REFER termination. You must upload an audio file (in .raw format) to /code/media onto the ESBC for the media you want played during the transfer. A separate file is required for each different codec type, even if the media itself is the same.

The playback configuration is defined listing the media files that you want to play. The playback-config element is configured under media-manager.

```
playback-config
  name                transferrbt
  entry
    encoding           PCMU
    filename            US_ringbackPCMU.raw
    bytes-per-sec      8000
```

The playback options can be applied to realms, sip-interfaces or session agents using the `spl-options` command.

Realm which requires media playback on REFER termination needs to have the following spl-options enabled

```
oraclesbc1(sip-interface)# spl-options playback-on-refer="transferrbt"
oraclesbc1(sip-interface)# done
```


Session Agents

Path: `configure terminal > session-router > session-agent`

Session agent	Transport	Enabled features
10.64.1.72 (PSTN)	TCP	OPTIONS
10.80.18.3 (CUCM cluster)	TCP	OPTIONS
10.89.17.7 (Avaya Aura)	TCP	OPTIONS
172.16.29.71 (SFB)	TCP	OPTIONS
med1.lyncclabsram.local(Lync 2013)	TLS	OPTIONS REFER termination
mediationserver.tekvizionlabs.local (CCE – SFB)	TLS	OPTIONS REFER Termination Delete +1 from RequestURI/TO/FROM headers

```

session-agent
  hostname                10.64.1.72
  ip-address              10.64.1.72
  transport-method       StaticTCP
  realm-id               PSTN
  description            PSTN
  ping-method            OPTIONS;hops=0
  ping-interval          30
session-agent
  hostname                10.80.18.3
  ip-address              10.80.18.3
  transport-method       StaticTCP
  realm-id               CUCMRealm
  description            CUCM
  ping-method            OPTIONS;hops=0
  ping-interval          30
session-agent
  hostname                10.89.17.7
  ip-address              10.89.17.7
  transport-method       StaticTCP
  realm-id               AvayaRealm
  description            Avaya
  ping-method            OPTIONS;hops=0
  ping-interval          30
session-agent
  hostname                172.16.29.71
  ip-address              172.16.29.71
  transport-method       StaticTCP
  realm-id               toSFB
  ping-method            OPTIONS
  ping-interval          30
session-agent

```

```

hostname                medi1.lyncclabsram.local
ip-address              172.16.31.97
port                   5067
transport-method       StaticTLS
realm-id               inside
ping-method            OPTIONS;hops=0
ping-interval          30
refer-call-transfer    enabled
session-agent
hostname                mediationserver.tekvizionlabs.local
ip-address              10.64.4.59
port                   5067
transport-method       StaticTLS
realm-id               outside
ping-method            OPTIONS;hops=0
ping-interval          30
in-translationid      delplus1
refer-call-transfer    enabled

```

Session Translation

Path: **configure terminal > session-router > session-translation**

```

session-translation
  id                    addPlus
  rules-calling         addPlus1
  rules-called          addPlus1
session-translation
  id                    delplus1
  rules-calling         delPlus1
  rules-called          delPlus1

```

SIP Config

Path: **configure terminal > session-router > sip-config > select**

```

sip-config
  home-realm-id        inside
  registrar-domain     *
  registrar-host       *
  registrar-port       5060
  options               max-udp-length=0

```

SIP Interfaces

Path: **configure terminal > session-router > sip-interface**

SIP-Interface	IP-Address	Transport	Optional features
AvayaRealm	10.70.50.21	TCP	<ul style="list-style-type: none"> Configure NatIpAvaya as out-manipulation Change 183 session progress to 180 Ringing
CUCMRealm	10.70.50.22	TCP	<ul style="list-style-type: none"> Enabled Secured-Network to allow calls into this realm with SRTP while not using TLS signaling Configured NatIp as out-manipulation

PSTN	10.64.4.145	TCP	<ul style="list-style-type: none"> Enabled Playback feature while terminating REFER Apply Natlp as out-manipulation
Inside	10.70.50.20	TCP & TLS	<ul style="list-style-type: none"> Enabled Secured-Network to allow calls into this realm with SRTP while not using TLS signaling Enabled Playback feature while terminating REFER Apply Stripsdp183 as in-manipulation Apply NatlpLync as out-manipulation
Outside	10.64.4.149	TCP&TLS	<ul style="list-style-type: none"> Apply Stripsdp183 as in-manipulation Apply NatlpCce as out-manipulation
outside-rtp	10.64.4.147	TCP	
toSFB	10.70.50.23	TCP	

```

sip-interface
  realm-id AvayaRealm
  sip-port
    address 10.70.50.21
    transport-protocol TCP
    allow-anonymous agents-only
  out-manipulationid NAT_IP_Avaya
  response-map change183to180
sip-interface
  realm-id CUCMRealm
  sip-port
    address 10.70.50.22
    transport-protocol TCP
    allow-anonymous agents-only
  secured-network enabled
  out-manipulationid NAT_IP
sip-interface
  realm-id PSTN
  sip-port
    address 10.64.4.145
    transport-protocol TCP
    allow-anonymous agents-only
  spl-options playback-on-refer="transferrbt"
  out-manipulationid NAT_IP
sip-interface
  realm-id inside
  sip-port
    address 10.70.50.20
    transport-protocol TCP
    allow-anonymous agents-only
  sip-port
    address 10.70.50.20
    port 5061

```

transport-protocol	TLS
tls-profile	LyncTLS
allow-anonymous	agents-only
secured-network	enabled
spl-options	playback-on-refer="transferrbt"
in-manipulationid	Stripsdp183
out-manipulationid	NatIpLync
sip-interface	
realm-id	outside
sip-port	
address	10.64.4.149
transport-protocol	TCP
allow-anonymous	agents-only
sip-port	
address	10.64.4.149
port	5061
transport-protocol	TLS
tls-profile	CCETLS
allow-anonymous	agents-only
in-manipulationid	Stripsdp183
out-manipulationid	NatIpCce
sip-interface	
realm-id	outside-rtsp
sip-port	
address	10.64.4.147
transport-protocol	TCP
sip-interface	
realm-id	toSFB
sip-port	
address	10.70.50.23
transport-protocol	TCP
allow-anonymous	agents-only

SIP Manipulations (Header Manipulation Rules – HMR)

Path: **configure terminal > session-router > sip-manipulation**

SIP manipulation (HMR)	Purpose
Changeinactosendonly	<ul style="list-style-type: none"> Following HMR changes inactive attribute in SDP to sendonly
NatIp	<ul style="list-style-type: none"> NAT From/To/Request-URI/PAI headers to contain appropriate uri-host info & Delete ms-source header from the INVITE
NatIpAvaya	<ul style="list-style-type: none"> NAT From/To/Request-URI/PAI headers to contain appropriate uri-host PAI header required to contain specific info – had to delete and overwrite contents form “From” header
NatIpCce	<ul style="list-style-type: none"> NAT From/To/Request-URI/PAI/Contact headers to contain appropriate uri-host Add +1 to refered by header Check if Request-uri/From/To contain 10 digits – if they do – add +1 Delete phone-context parameter from “From” header
NatIpLync	<ul style="list-style-type: none"> NAT From/To/Request-URI/PAI/Contact headers

Stripsdp183	<ul style="list-style-type: none"> • Check & delete SDP from 183 Session Progress message, delete SDP & change inactive to sendonly
alter4xxOPTIONS	<ul style="list-style-type: none"> • Change 483 & 404 responses OPTIONS to 200 OK
delReqNatIp	<ul style="list-style-type: none"> • Delete Require Header & Apply NatIp sip-manipulation
alterreferby	<ul style="list-style-type: none"> • NAT uri-host part of refered-By header
stripRR	<ul style="list-style-type: none"> • Delete any record-route headers

```

sip-manipulation
  name                               Changeinactosendonly
  description                         Change inactive to sendonly for pstn tran
  header-rule
    name                               changeSDP
    header-name                         Content-Type
    action                               manipulate
    msg-type                             request
    methods                              INVITE
    element-rule
      name                               inacttosendonly
      parameter-name                     application/sdp
      type                                mime
      action                              find-replace-all
      comparison-type                    pattern-rule
      match-value                        a=inactive
      new-value                           a=sendonly

sip-manipulation
  name                               NatIp
  header-rule
    name                               From
    header-name                         From
    action                               manipulate
    msg-type                             request
    element-rule
      name                               fromEr
      type                                uri-host
      action                              find-replace-all
      new-value                           oracle.tekvizionlabs.local

  header-rule
    name                               To
    header-name                         To
    action                               manipulate
    msg-type                             request
    element-rule
      name                               ToEr
      type                                uri-host
      action                              find-replace-all
      new-value                           $REMOTE_IP

  header-rule
    name                               RequestURI
    header-name                         request-uri
    action                               manipulate
    methods                              INVITE

```

```

        element-rule
            name                ruriEr
            type                uri-host
            action              find-replace-all
            new-value           $REMOTE_IP

    header-rule
        name                    alterPAI
        header-name             P-Asserted-Identity
        action                  manipulate
        element-rule
            name                alterPAIEr
            type                uri-host
            action              find-replace-all
            new-value           $LOCAL_IP

    header-rule
        name                    msSource
        header-name             ms-call-source
        action                  delete
        methods                 INVITE

sip-manipulation
    name                       NatIpAvaya
    header-rule
        name                    From
        header-name             From
        action                  manipulate
        msg-type                request
        element-rule
            name                fromEr
            type                uri-host
            action              find-replace-all
            new-value           oracle.tekvizionlabs.local

        element-rule
            name                store_user
            type                uri-user
            action              store

    header-rule
        name                    To
        header-name             To
        action                  manipulate
        msg-type                request
        element-rule
            name                ToEr
            type                uri-host
            action              find-replace-all
            new-value           $REMOTE_IP

    header-rule
        name                    RequestURI
        header-name             request-uri
        action                  manipulate
        methods                 INVITE
        element-rule
            name                ruriEr
            type                uri-host
            action              find-replace-all
            new-value           $REMOTE_IP

    header-rule

```

```

        name                delPAI
        header-name         P-Asserted-Identity
        action              delete
        msg-type            request
        methods             INVITE
header-rule
    name                    msSource
    header-name             ms-call-source
    action                  delete
    methods                 INVITE
header-rule
    name                    createPAI
    header-name             Contact
    action                  manipulate
    msg-type                request
    methods                 INVITE
    match-value
    element-rule
        name                updateContact
        type                 uri-user
        action               find-replace-all
        new-value            $From.$store_user.$0
header-rule
    name                    storeContact
    header-name             Contact
    action                  store
    comparison-type         pattern-rule
    msg-type                request
    methods                 INVITE
header-rule
    name                    addPAI
    header-name             P-Asserted-Identity
    action                  add
    comparison-type         boolean
    msg-type                request
    methods                 INVITE
    match-value             $delPAI
    new-value                $storeContact.$0
sip-manipulation
    name                    NatIpCce
    header-rule
        name                From
        header-name         From
        action              manipulate
        methods             INVITE
        element-rule
            name                fromEr
            type                 uri-host
            action               find-replace-all
            new-value            oracle.tekvizionlabs.local
    header-rule
        name                To
        header-name         To
        action              manipulate
        methods             INVITE
        element-rule

```

```

        name ToEr
        type uri-host
        action find-replace-all
        new-value
mediationserver.tekvizionlabs.local
  header-rule
    name RequestURI
    header-name request-uri
    action manipulate
    methods INVITE
    element-rule
      name ruriEr
      type uri-host
      action find-replace-all
      new-value
mediationserver.tekvizionlabs.local
  header-rule
    name alterPAI
    header-name P-Asserted-Identity
    action manipulate
    element-rule
      name alterPAIEr
      type uri-host
      action find-replace-all
      new-value $LOCAL_IP
  header-rule
    name Contact
    header-name Contact
    action manipulate
    methods INVITE
    element-rule
      name contactEr
      type uri-host
      action find-replace-all
      new-value oracle.tekvizionlabs.local
  header-rule
    name AddPlus1
    header-name referred-by
    action manipulate
    element-rule
      name addplus1
      type uri-user
      action find-replace-all
      new-value "+1"+$ORIGINAL
  header-rule
    name alterreferby
    header-name From
    action sip-manip
    msg-type request
    new-value alterreferby
  header-rule
    name removePC
    header-name From
    action manipulate
    methods INVITE
    element-rule

```



```

        name removePCEr
        parameter-name phone-context
        type uri-user-param
        action delete-element

header-rule
    name
    header-name FromUser
    action from
    methods manipulate
    element-rule INVITE
        name fromEr
        type uri-user
        action find-replace-all
        match-value ^(\d{10})
        new-value "+1"+$ORIGINAL

header-rule
    name Plus1RURI
    header-name Request-URI
    action manipulate
    element-rule
        name PlusRURIEr
        type uri-user
        action find-replace-all
        match-value ^(\d{10})
        new-value "+1"+$ORIGINAL
    element-rule
        name PlusRURI_store
        type uri-user
        action store

header-rule
    name Plus1To
    header-name To
    action manipulate
    element-rule
        name PlusToEr
        type uri-user
        action find-replace-all
        match-value ^(\d{10})
        new-value "+1"+$ORIGINAL
    element-rule
        name copyRURI
        type uri-user
        action find-replace-all
        match-value ^(\d{4})
        new-value

$Plus1RURI.$PlusRURI_store.$0
sip-manipulation
    name NatIpLync
    header-rule
        name from
        header-name From
        action sip-manip
        new-value NAT_IP
    header-rule
        name alterPAI
        header-name P-Asserted-Identity

```

```

        action                manipulate
        methods                INVITE
        element-rule
            name                alterPAIEr
            type                uri-host
            action              find-replace-all
            new-value           oracle.tekvizionlabs.local
    header-rule
        name                    alterContact
        header-name             Contact
        action                  manipulate
        methods                 INVITE
        element-rule
            name                alterContactEr
            type                uri-host
            action              find-replace-all
            new-value           oracle.tekvizionlabs.local
    header-rule
        name                    To
        header-name             To
        action                  manipulate
        methods                 INVITE
        element-rule
            name                toEr
            type                uri-host
            action              find-replace-all
            new-value           med1.lyncclabsram.local
    header-rule
        name                    ruri
        header-name             request-uri
        action                  manipulate
        methods                 INVITE
        element-rule
            name                ruriEr
            type                uri-host
            action              find-replace-all
            new-value           med1.lyncclabsram.local
    header-rule
        name                    alterReferredBy
        header-name             Referred-By
        action                  manipulate
        msg-type                request
        methods                 INVITE
        element-rule
            name                alterRB
            type                uri-host
            action              find-replace-all
            new-value           oracle.tekvizionlabs.local
sip-manipulation
    name                        Stripsdp183
    description                 For incoming 183 from Lync, strip SDP
    header-rule
        name                    check183
        header-name             @status-line
        action                  store
        comparison-type         pattern-rule

```

```

        element-rule
            name                is183
            type                 status-code
            action                store
            comparison-type       pattern-rule
            match-value           183
header-rule
    name                delSDP
    header-name         Content-Type
    action              manipulate
    comparison-type     case-insensitive
    match-value         $check183.$is183
    element-rule
        name                del183SDP
        parameter-name       application/sdp
        type                 mime
        action                delete-element
        comparison-type       boolean
header-rule
    name                delContentType
    header-name         Content-Type
    action              manipulate
    comparison-type     boolean
    match-value         $check183.$is183
    element-rule
        name                delCT
        parameter-name       *
        type                 header-param
        action                delete-header
header-rule
    name                inactosendonly
    header-name         From
    action              sip-manip
    msg-type             request
    new-value           Changeinactosendonly
sip-manipulation
    name                alter4xxOPTIONS
header-rule
    name                alterRURI
    header-name         Request-URI
    action              manipulate
    msg-type             reply
    methods              OPTIONS
    element-rule
        name                alterRURIStatusCode
        type                 status-code
        action                find-replace-all
        match-value           483 || 404
        new-value             200
    element-rule
        name                alterRURIReasonPhrase
        type                 reason-phrase
        action                find-replace-all
        match-value           Too Many Hops || Not Found
        new-value             OK

```

```

sip-manipulation
  name                               delReqNatIp
  header-rule
    name                               delRequire
    header-name                         Require
    action                               delete
    msg-type                             reply
    methods                              INVITE
  header-rule
    name                               nat
    header-name                         From
    action                               sip-manip
    new-value                            NAT_IP
sip-manipulation
  name                               alterreferby
  header-rule
    name                               checkReferBy
    header-name                         REFERRED-BY
    action                               manipulate
  element-rule
    name                               checkReferBy
    type                               uri-host
    action                               find-replace-all
    new-value                            oracle.tekvizionlabs.local
sip-manipulation
  name                               stripRR
  join-headers                         Record-Route
  header-rule
    name                               delRR
    header-name                         Record-Route
    action                               delete

```

SIP Monitoring

Path: **configure terminal > session-router > sip-monitoring > select**

```

sip-monitoring
  match-any-filter                     enabled
  monitoring-filters                    all

```

Steering Pools

Path: **configure terminal > media-manager > steering-pool**

```

steering-pool
  ip-address                           10.64.4.145
  start-port                            49600
  end-port                               65535
  realm-id                              PSTN
steering-pool
  ip-address                           10.64.4.147
  start-port                            49600
  end-port                               65535
  realm-id                              outside-rtp
steering-pool

```

ip-address	10.64.4.149
start-port	49600
end-port	65535
realm-id	outside
steering-pool	
ip-address	10.70.50.20
start-port	49562
end-port	65535
realm-id	inside
steering-pool	
ip-address	10.70.50.21
start-port	49600
end-port	65535
realm-id	AvayaRealm
steering-pool	
ip-address	10.70.50.22
start-port	49600
end-port	65535
realm-id	CUCMRealm
steering-pool	
ip-address	10.70.50.23
start-port	49562
end-port	65535
realm-id	toSFB

Certificate record

Path: **configure terminal > security > certificate-record**

certificate-record	
name	LyncMed
state	TX
locality	Plano
common-name	medserver1.lyncclabsram.local
digest-algor	sha1
certificate-record	
name	LyncRoot
state	TX
locality	Plano
common-name	lyncclabsram-DC-CA
digest-algor	sha1
certificate-record	
name	SBCCE2
state	TX
locality	Plano
common-name	oracle.tekvizionlabs.local
digest-algor	sha1
certificate-record	
name	SBCCert
state	TX
locality	Plano
common-name	oracle.tekvizionlabs.local
digest-algor	sha1
certificate-record	
name	SBCCertCCE

```

state TX
locality Plano
common-name oracle.tekvizionlabs.local
digest-algor sha1
certificate-record
name cceRoot
state TX
locality Plano
common-name SfB CCE Root
digest-algor sha1
certificate-record
name newCCERoot
state TX
locality Plano
common-name SfB CCE Root
digest-algor sha1
certificate-record
name rootcert
state TX
locality Plano
common-name lynclabsram-DC-CA
digest-algor sha1

```

TLS profile

Path: **configure terminal > security > tls-profile**

Following two tls-profiles are required - one towards CCE while the other towards Lync 2013 - as both CCE & Lync 2013 use different root certs

```

tls-profile
name CCETLS
end-entity-certificate SBCCCE2
trusted-ca-certificates newCCERoot
mutual-authenticate enabled
tls-version tlsv12
tls-profile
name LyncTLS
end-entity-certificate SBCCert
trusted-ca-certificates rootcert
mutual-authenticate LyncRoot
mutual-authenticate enabled
tls-version tlsv1

```

Tls Global config

Path: **configure terminal > security > tls-global**

The following config element needs to be enabled in order for the SBC to cache TLS connections

```

tls-global
session-caching enabled

```

Sdes Profile

Path: **configure terminal > security > media-security > sdes-profile**

sdes-profile		
name		SRTP
use-ingress-session-params		srtcp-encrypt

Media-sec-policy

Path: **configure terminal > security > media-security > media-sec-policy**

media-sec-policy			
name			RTP
media-sec-policy			
name			SRTP
inbound			
profile			SRTP
mode			srtcp
protocol			sdes
outbound			
profile			SRTP
mode			srtcp
protocol			sdes

RTCP Policy

Path: **configure terminal > media-manager > rtcp-policy**

rtcp-policy		
name		rtcpGen
rtcp-generate		all-calls

Response Map

Path: **configure terminal > session-router > sip-response-map**

response-map		
name		change183to180
entries		
recv-code		183
xmit-code		180
reason		Ringing

LDAP Config

Path: `configure terminal > session-router > ldap-config`

```

ldap-config
  name                    ldapCfg
  ldap-servers            172.16.31.91:389
  realm                   inside
  username                administrator@lynclabsram.local
  password                *****
  ldap-search-base       CN=Users,DC=lynclabsram,DC=local
  ldap-transactions
    route-mode            attribute-order-only
    ldap-cfg-attributes
      name                 msRTCSIP-Line
      next-hop             med1.lynclabsram.local
      realm                inside
      extraction-regex     (.*)
      value-format         tel:$0
    ldap-cfg-attributes
      name                 telephoneNumber
      next-hop             mediationserver.tekvizionlabs.local
  realm                   outside
  extraction-regex        (.*)
  value-format            tel:$0
  
```

Codec-policy

Path: `configure terminal > media-manager > codec-policy`

Codec-policy	Purpose
G711Only	Allow only G711 & telephone-event, remove G729
G729only	Allow G729 but remove G711 codecs
Xcode	Allow all codecs, but add comfort noise (payload 13)
toSFB	Allow all codecs, but add comfort noise (payload 13)

```

codec-policy
  name                    G711Only
  allow-codecs            * G729:no
  add-codecs-on-egress   PCMU PCMA telephone-event
codec-policy
  name                    G729only
  allow-codecs            * PCMA:no PCMU:no
  add-codecs-on-egress   G729
codec-policy
  name                    Xcode
  allow-codecs            *
  add-codecs-on-egress   CN
  
```



```

codec-policy
  name                toSFB
  allow-codecs        *
  add-codecs-on-egress  CN

```

System Config

Path: **configure terminal > system > system-config > select**

```

system-config
  process-log-level    WARNING
  default-gateway     192.65.79.97

```

Translation Rules

Path: **configure terminal > session-router > translation-rule**

```

translation-rules
  id                addPlus1
  type              add
  add-string        +1
translation-rules
  id                delPlus1
  type              delete
  delete-string     +1

```

Web Server Config

Path: **configure terminal > system > web-server-config > select**

```

web-server-config
  state              enabled
  inactivity-timeout 5
  http-state         enabled
  http-port          80
  https-state        disabled
  https-port         443
  tls-profile

```


Save, Activate, and Reboot

You will now save your configuration with the **save-config** command. This will make it persistent through reboots, but it will not take effect until after you issue the **activate-config** command. Some config elements are not Real-Time Configuration (RTC) supported, so a reboot is required after the initial configuration.

```

oraclesbcl# save-config
checking configuration
Save-Config received, processing.
waiting 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'.
oraclesbc1# activate-config
Activate-Config received, processing.
waiting for request to finish
Setting phy0 on Slot=0, Port=0, MAC=00:08:25:03:FC:43, VMAC=00:08:25:03:FC:43
Setting phy1 on Slot=1, Port=0, MAC=00:08:25:03:FC:45, VMAC=00:08:25:03:FC:45
Request to 'ACTIVATE-CONFIG' has Finished,
Activate Complete
oraclesbc1# reboot force
```

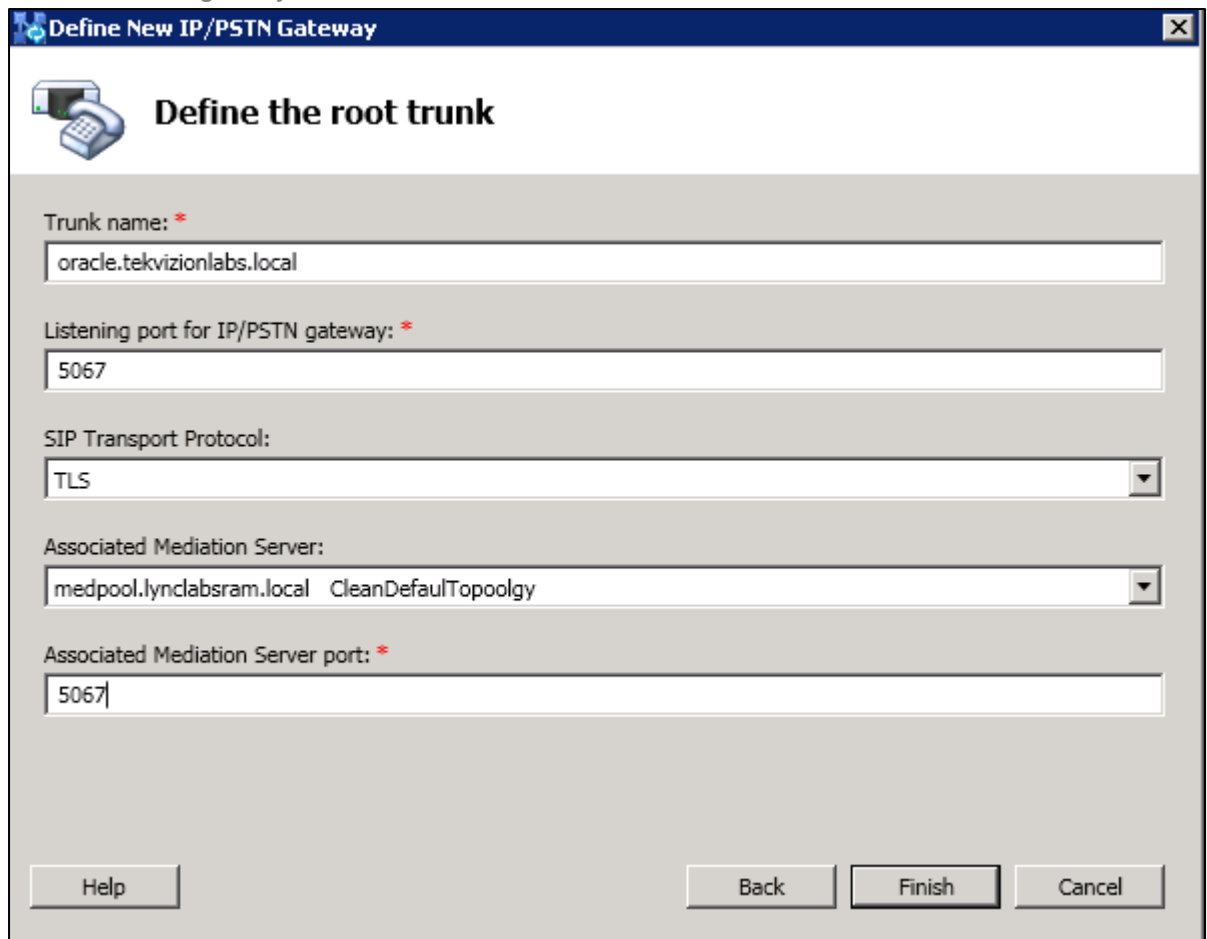
The E-SBC configuration is now complete.

Sections below will walk thru configuring all of the 3rd party products and how they need to be configured in order for them to communicate with Oracle ESBC.

- Lync 2013
- Avaya Session Manager
- Cisco UCM
- Microsoft CCE

Lync 2013(10.70.50.20)

Topology builder with PSTN gateway



Define New IP/PSTN Gateway

Define the root trunk

Trunk name: *
oracle.tekvizionlabs.local

Listening port for IP/PSTN gateway: *
5067

SIP Transport Protocol:
TLS

Associated Mediation Server:
medpool.lyncclabsram.local CleanDefaultTopoolgy

Associated Mediation Server port: *
5067

Help Back Finish Cancel

Figure 1: Defining the Root Trunk Details

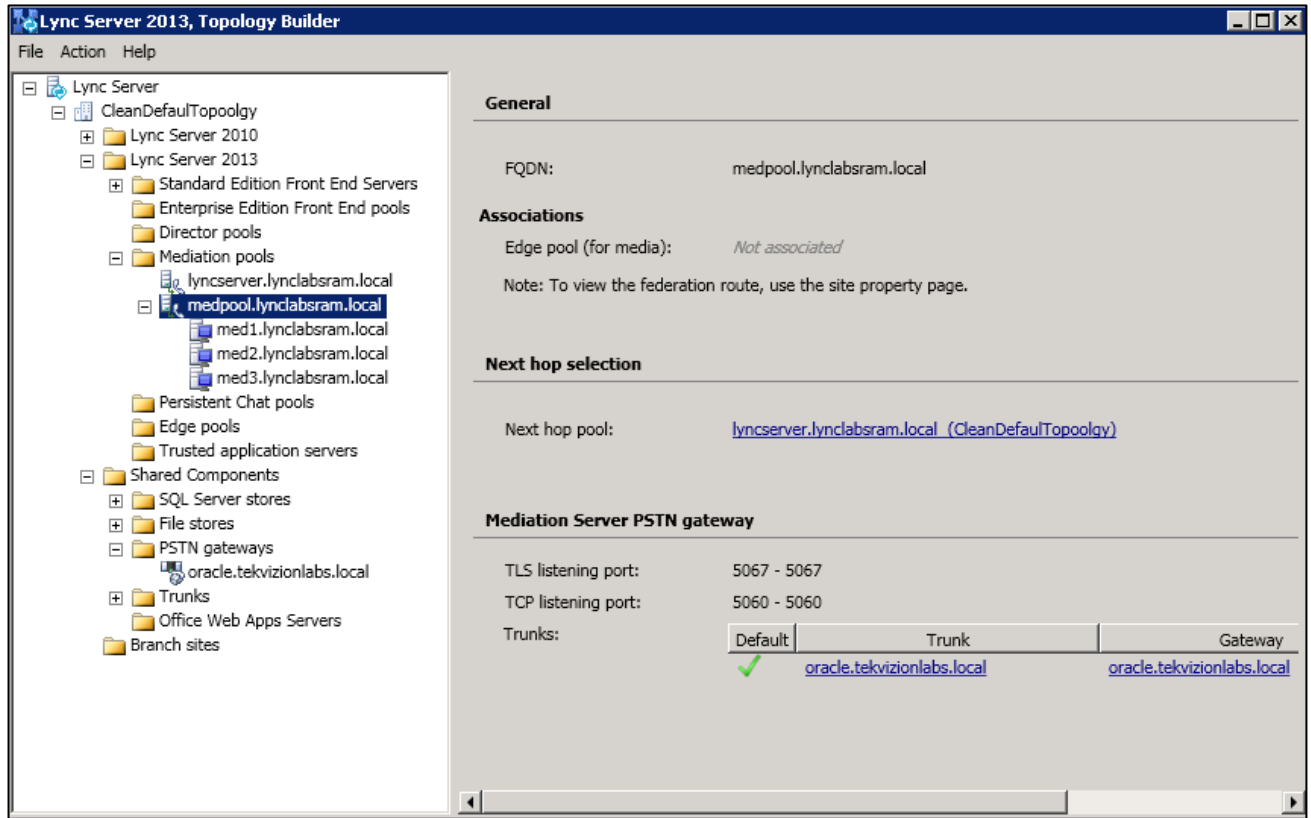


Figure 2: Mediation server with Associated PSTN gateway

Lync 2013 Control panel with "Associated PSTN gateway"

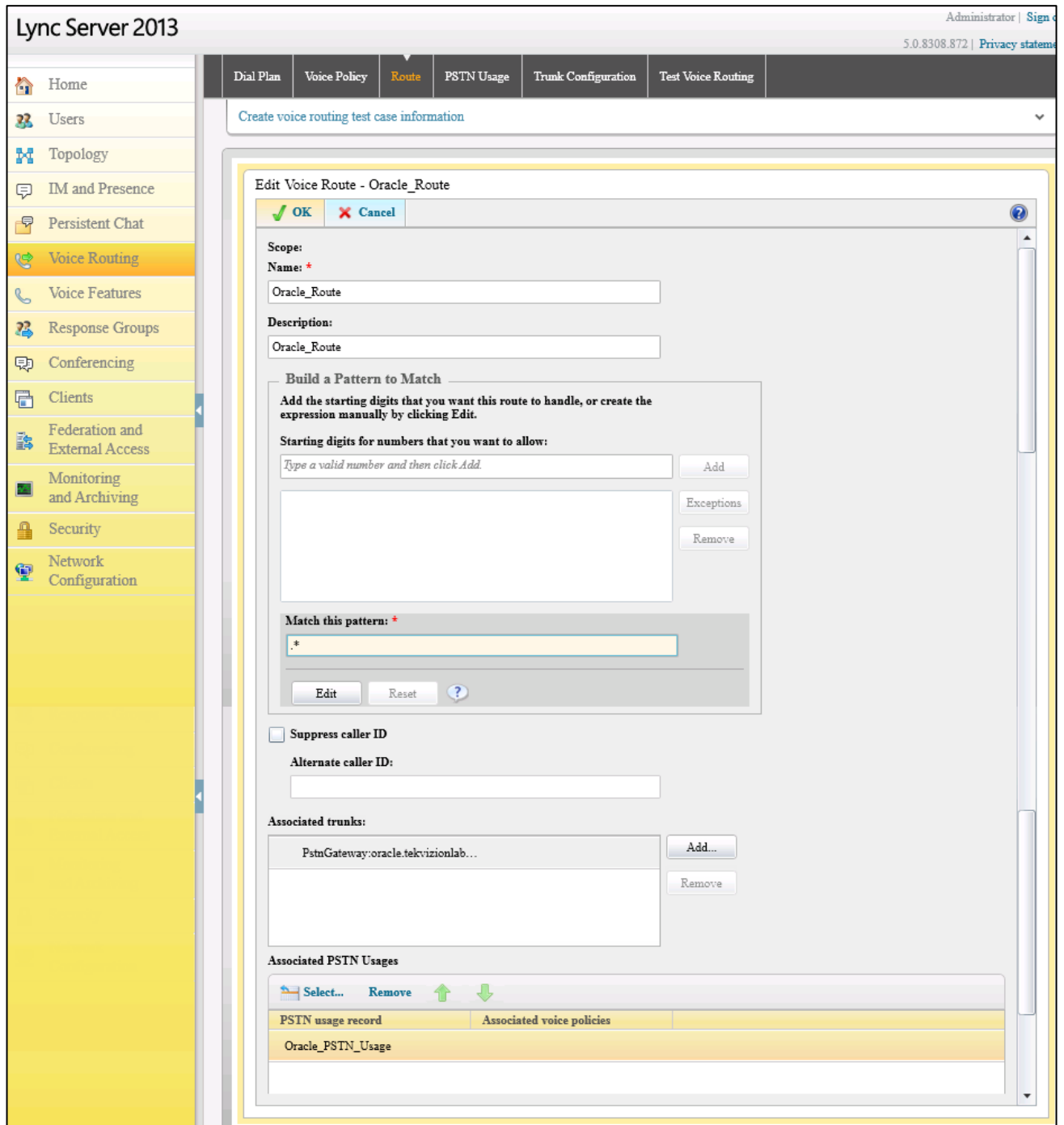


Figure 3: Associating PSTN Gateway to newly created Route

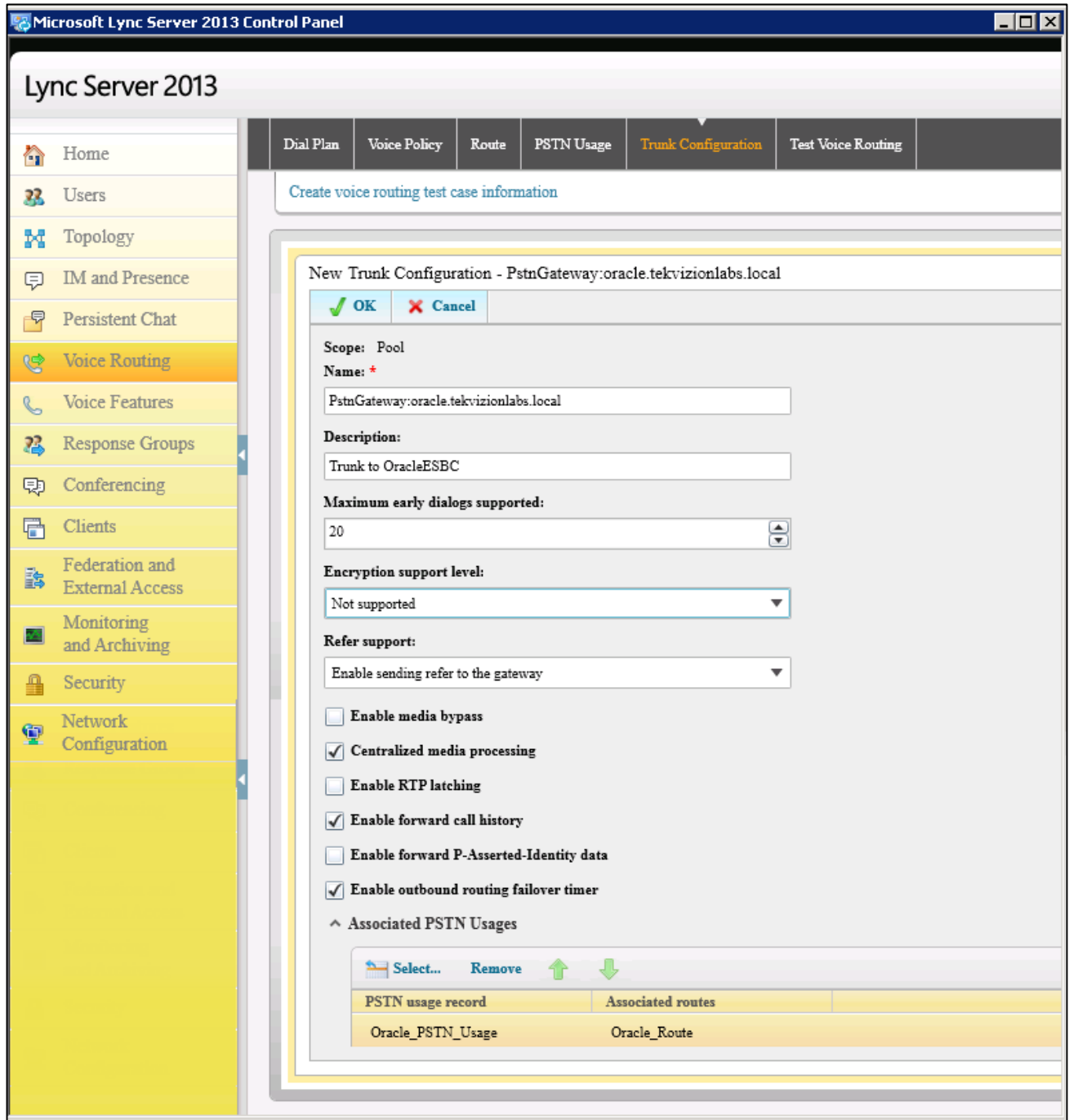


Figure 4: Trunk Configuration

Avaya (10.70.50.21)

SBC Configured on Session Manger

SIP Entity Configured on Session Manger for Oracle ESBC

The screenshot displays the Avaya Aura System Manager 7.0 interface. The top navigation bar includes 'Home' and 'Routing' tabs. A left-hand menu lists various configuration options: Domains, Locations, Adaptations, SIP Entities (selected), Entity Links, Time Ranges, Routing Policies, Dial Patterns, Regular Expressions, and Defaults. The main content area is titled 'SIP Entity Details' and is divided into three sections: General, Loop Detection, and SIP Link Monitoring. The 'General' section contains fields for Name (OracleEsbc), FQDN or IP Address (10.70.50.21), Type (Other), Notes, Adaptation (CallsToOracleSBC), Location (Plano), Time Zone (America/Fortaleza), SIP Timer B/F (4), Credential name, Securable (unchecked), Call Detail Recording (none), and CommProfile Type Preference. The 'Loop Detection' section includes Loop Detection Mode (On), Loop Count Threshold (5), and Loop Detection Interval (200). The 'SIP Link Monitoring' section features SIP Link Monitoring (Link Monitoring Enabled), Proactive Monitoring Interval (120), Reactive Monitoring Interval (60), Number of Tries (1), Supports Call Admission Control (unchecked), Shared Bandwidth Manager (unchecked), Primary Session Manager Bandwidth Association, and Backup Session Manager Bandwidth Association. The interface also shows a breadcrumb trail 'Home / Elements / Routing / SIP Entities' and 'Commit'/'Cancel' buttons.

SIP Entity Details

General

* Name: OracleEsbc

* FQDN or IP Address: 10.70.50.21

Type: Other

Notes:

Adaptation: CallsToOracleSBC

Location: Plano

Time Zone: America/Fortaleza

* SIP Timer B/F (in seconds): 4

Credential name:

Securable:

Call Detail Recording: none

CommProfile Type Preference:

Loop Detection

Loop Detection Mode: On

Loop Count Threshold: 5

Loop Detection Interval (in msec): 200

SIP Link Monitoring

SIP Link Monitoring: Link Monitoring Enabled

* Proactive Monitoring Interval (in seconds): 120

* Reactive Monitoring Interval (in seconds): 60

* Number of Tries: 1

Supports Call Admission Control:

Shared Bandwidth Manager:

Primary Session Manager Bandwidth Association:

Backup Session Manager Bandwidth Association:

Figure 5: SIP Entity for OracleESBC

Entity Link configured on Session Manger for Oracle ESBC

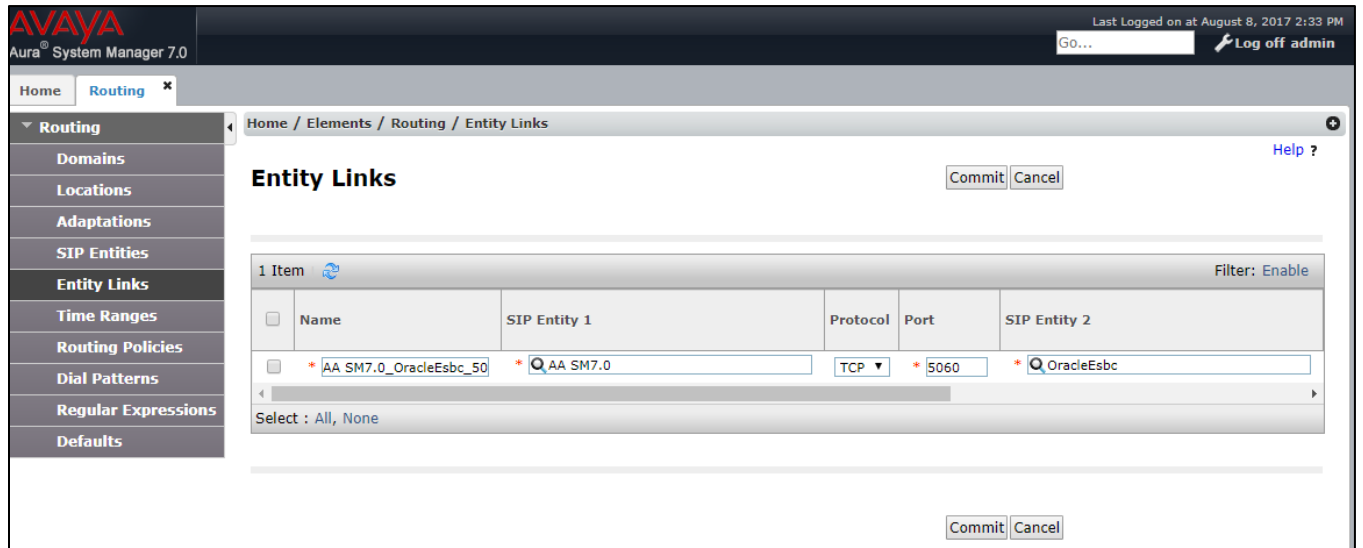


Figure 6: Entity Link to Oracle ESBC

Routing Policy

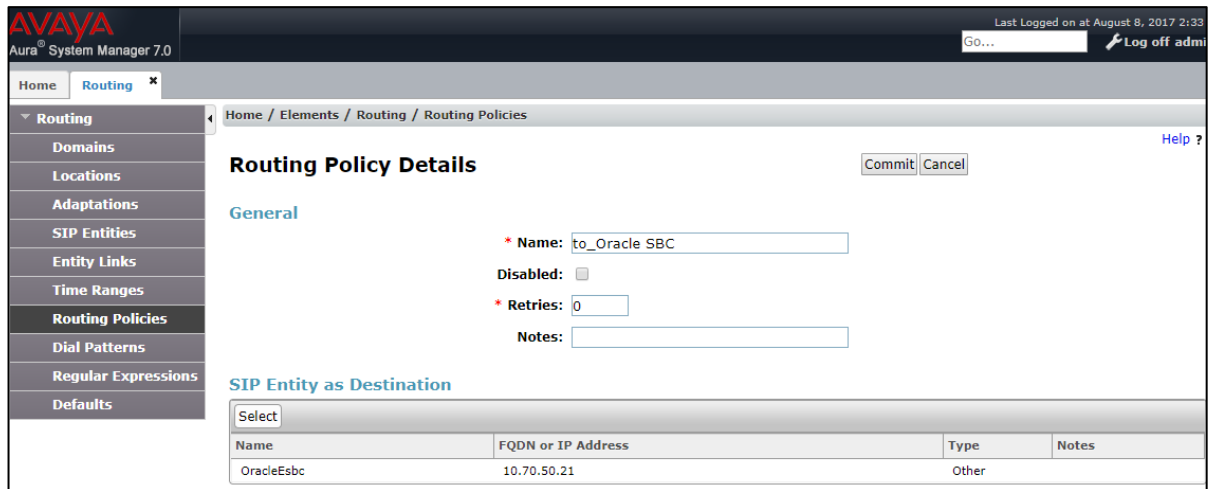


Figure 7: Routing Policy for Oracle ESBC

Dial Pattern

Home / Elements / Routing / Dial Patterns

Dial Pattern Details

Commit Cancel

General

* Pattern:

* Min:

* Max:

Emergency Call:

Emergency Priority:

Emergency Type:

SIP Domain:

Notes:

Originating Locations and Routing Policies

Add Remove

1 Item Filter: Enable

<input type="checkbox"/>	Originating Location Name	Originating Location Notes	Routing Policy Name	Rank	Routing Policy Disabled	Routing Policy Destination	Routing Policy Notes
<input type="checkbox"/>	Plano	Avaya Aura 7.0	to_Oracle SBC	0	<input type="checkbox"/>	OracleEsbcc	

Select : All, None

Figure 8: Dial Pattern for 10 Digit dialing

Home / Elements / Routing / Dial Patterns

Dial Pattern Details

Commit Cancel

General

* Pattern:

* Min:

* Max:

Emergency Call:

Emergency Priority:

Emergency Type:

SIP Domain:

Notes:

Originating Locations and Routing Policies

Add Remove

1 Item Filter: Enable

<input type="checkbox"/>	Originating Location Name	Originating Location Notes	Routing Policy Name	Rank	Routing Policy Disabled	Routing Policy Destination	Routing Policy Notes
<input type="checkbox"/>	Plano	Avaya Aura 7.0	to_Oracle SBC	0	<input type="checkbox"/>	OracleEsbcc	

Select : All, None

Figure 9: Dial Pattern for Extension dialing

Cisco UCM (10.70.50.22)

SIP Trunk List



Trunks (1 - 2 of 2)		Rows per Page 50											
Find Trunks where		Device Name	begins with	Find		Clear Filter							
Select item or enter search text													
<input type="checkbox"/>		Name ^	Description	Calling Search Space	Device Pool	Route Pattern	Partition	Route Group	Priority	Trunk Type	SIP Trunk Status	SIP Trunk Duration	SIP Trunk Security Profile
<input type="checkbox"/>		Trunk_to_OracleSBC	Trunk_to_OracleSBC		Oracle Device Pool	2.XXXX				SIP Trunk	Full Service	Time In Full Service: 1 day 14 hours 7 minutes	Oracle Non Secure SIP Trunk Profile
<input type="checkbox"/>		Trunk_to_OracleSBC	Trunk_to_OracleSBC		Oracle Device Pool	3.XXXXXXXXXX				SIP Trunk	Full Service	Time In Full Service: 1 day 14 hours 7 minutes	Oracle Non Secure SIP Trunk Profile

Figure 10: SIP Trunk List

SIP Trunk Configuration

SIP Trunk Status	
Service Status:	Full Service
Duration:	Time In Full Service: 1 day 14 hours 24 minutes
Device Information	
Product:	SIP Trunk
Device Protocol:	SIP
Trunk Service Type	None(Default)
Device Name*	<input type="text" value="Trunk_to_OracleSBC"/>
Description	<input type="text" value="Trunk_to_OracleSBC"/>
Device Pool*	<input type="text" value="Oracle Device Pool"/>
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="MRGL_Default"/>
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"/>

Figure 11 : SIP Trunk to OracleESBC

Media Termination Point Required
 Retry Video Call as Audio
 Path Replacement Support
 Transmit UTF-8 for Calling Party Name
 Transmit UTF-8 Names in QSIG APDU
 Unattended Port
 SRTP Allowed - When this flag is checked, Encrypted TLS needs to be configured in the network to provide end to end security. Failure to do so will expose keys and other information.
 Consider Traffic on This Trunk Secure*
 Route Class Signaling Enabled*
 Use Trusted Relay Point*
 PSTN Access
 Run On All Active Unified CM Nodes

Intercompany Media Engine (IME)

E.164 Transformation Profile

MLPP and Confidential Access Level Information

MLPP Domain
 Confidential Access Mode
 Confidential Access Level

Call Routing Information

Remote-Party-Id
 Asserted-Identity
 Asserted-Type*
 SIP Privacy*

Inbound Calls

Significant Digits*
 Connected Line ID Presentation*
 Connected Name Presentation*
 Calling Search Space
 AAR Calling Search Space
 Prefix DN
 Redirecting Diversion Header Delivery - Inbound

Incoming Calling Party Settings

If the administrator sets the prefix to Default this indicates call processing will use prefix at the next level setting (DevicePool/Service Parameter). Otherwise, the value configured is used as the prefix unless the field is empty in which case there is no prefix assigned.

Number Type	Prefix	Strip Digits	Calling Search Space	Use Device Pool CSS
Incoming Number	<input type="text" value="Default"/>	<input type="text" value="0"/>	<input type="text" value="< None >"/>	<input checked="" type="checkbox"/>

Incoming Called Party Settings

If the administrator sets the prefix to Default this indicates call processing will use prefix at the next level setting (DevicePool/Service Parameter). Otherwise, the value configured is used as the prefix unless the field is empty in which case there is no prefix assigned.

Number Type	Prefix	Strip Digits	Calling Search Space	Use Device Pool CSS
Incoming Number	<input type="text" value="Default"/>	<input type="text" value="0"/>	<input type="text" value="< None >"/>	<input checked="" type="checkbox"/>

Figure 12: SIP Trunk to OracleESBC (Cont.)

Connected Party Settings

Connected Party Transformation CSS < None >

Use Device Pool Connected Party Transformation CSS

Outbound Calls

Called Party Transformation CSS < None >

Use Device Pool Called Party Transformation CSS

Calling Party Transformation CSS < None >

Use Device Pool Calling Party Transformation CSS

Calling Party Selection* Originator

Calling Line ID Presentation* Default

Calling Name Presentation* Default

Calling and Connected Party Info Format* Deliver DN only in connected party

Redirecting Diversion Header Delivery - Outbound

Redirecting Party Transformation CSS < None >

Use Device Pool Redirecting Party Transformation CSS

Caller Information

Caller ID DN

Caller Name

Maintain Original Caller ID DN and Caller Name in Identity Headers

SIP Information

Destination

Destination Address is an SRV

	Destination Address	Destination Address IPv6	Destination Port
1*	<input type="text" value="10.70.50.22"/>	<input type="text"/>	<input type="text" value="5060"/>

MTP Preferred Originating Codec* 711ulaw

BLF Presence Group* Standard Presence group

SIP Trunk Security Profile* Oracle Non Secure SIP Trunk Profile

Rerouting Calling Search Space < None >

Out-Of-Dialog Refer Calling Search Space < None >

SUBSCRIBE Calling Search Space < None >

SIP Profile* Oracle_Standard SIP Profile [View Details](#)

DTMF Signaling Method* No Preference

Normalization Script

Normalization Script < None >

Enable Trace

	Parameter Name	Parameter Value
1	<input type="text"/>	<input type="text"/> <input type="button" value="+"/> <input type="button" value="-"/>

Recording Information

None

This trunk connects to a recording-enabled gateway

This trunk connects to other clusters with recording-enabled gateways

Geolocation Configuration

Geolocation < None >

Geolocation Filter < None >

Send Geolocation Information

Figure 13 : SIP Trunk to OracleESBC (Cont.)

Route Pattern





Route Patterns (1 - 2 of 2)						Rows per Page 50
Find Route Patterns where						
Pattern		begins with		Find	Clear Filter	 
<input type="checkbox"/>	Pattern ^	Description	Partition	Route Filter	Associated Device	Copy
<input type="checkbox"/>	3.XXXXXXXXXX	Oracle 10 Digit Dialing			Trunk to OracleSBC	
<input type="checkbox"/>	2.XXXX	Oracle Extension Dialing			Trunk to OracleSBC	

Figure 14: Route Patterns List

Route Pattern for 10 Digit Dialing

Pattern Definition		
Route Pattern*	3.XXXXXXXXXX	
Route Partition	< None >	
Description	Oracle 10 Digit Dialing	
Numbering Plan	-- Not Selected --	
Route Filter	< None >	
MLPP Precedence*	Default	
<input type="checkbox"/> Apply Call Blocking Percentage		
Resource Priority Namespace Network Domain	< None >	
Route Class*	Default	
Gateway/Route List*	Trunk_to_OracleSBC	(Edit)
Route Option	<input checked="" type="radio"/> Route this pattern <input type="radio"/> Block this pattern No Error	
Call Classification*	OffNet	
External Call Control Profile	< None >	
<input type="checkbox"/> Allow Device Override	<input checked="" type="checkbox"/> Provide Outside Dial Tone	<input type="checkbox"/> Allow Overlap Sending
<input type="checkbox"/> Require Forced Authorization Code		
Authorization Level*	0	
<input type="checkbox"/> Require Client Matter Code		
Calling Party Transformations		
<input checked="" type="checkbox"/> Use Calling Party's External Phone Number Mask		
Calling Party Transform Mask		
Prefix Digits (Outgoing Calls)		
Calling Line ID Presentation*	Default	
Calling Name Presentation*	Default	
Calling Party Number Type*	Cisco CallManager	
Calling Party Numbering Plan*	Cisco CallManager	
Connected Party Transformations		
Connected Line ID Presentation*	Default	
Connected Name Presentation*	Default	
Called Party Transformations		
Discard Digits	PreDot	
Called Party Transform Mask		
Prefix Digits (Outgoing Calls)		
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		
Network Service	Service Parameter Name	Service Parameter Value
-- Not Selected --	< Not Exist >	
<input type="button" value="Save"/> <input type="button" value="Delete"/> <input type="button" value="Copy"/> <input type="button" value="Add New"/>		

Figure 15: Route Pattern for 10 Digit Dialing

Route Pattern for Extension Dialing

Pattern Definition		
Route Pattern*	2.XXXX	
Route Partition	< None >	
Description	Oracle Extension Dialing	
Numbering Plan	-- Not Selected --	
Route Filter	< None >	
MLPP Precedence*	Default	
<input type="checkbox"/> Apply Call Blocking Percentage		
Resource Priority Namespace Network Domain	< None >	
Route Class*	Default	
Gateway/Route List*	Trunk_to_OracleSBC	(Edit)
Route Option	<input checked="" type="radio"/> Route this pattern <input type="radio"/> Block this pattern No Error	
Call Classification*	OffNet	
External Call Control Profile	< None >	
<input type="checkbox"/> Allow Device Override	<input checked="" type="checkbox"/> Provide Outside Dial Tone	<input type="checkbox"/> Allow Overlap Sending
<input type="checkbox"/> Require Forced Authorization Code		
Authorization Level*	0	
<input type="checkbox"/> Require Client Matter Code		
Calling Party Transformations		
<input type="checkbox"/> Use Calling Party's External Phone Number Mask		
Calling Party Transform Mask		
Prefix Digits (Outgoing Calls)		
Calling Line ID Presentation*	Default	
Calling Name Presentation*	Default	
Calling Party Number Type*	Cisco CallManager	
Calling Party Numbering Plan*	Cisco CallManager	
Connected Party Transformations		
Connected Line ID Presentation*	Default	
Connected Name Presentation*	Default	
Called Party Transformations		
Discard Digits	PreDot	
Called Party Transform Mask		
Prefix Digits (Outgoing Calls)		
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		
Network Service	Service Parameter Name	Service Parameter Value
-- Not Selected --	< Not Exist >	
<input type="button" value="Save"/> <input type="button" value="Delete"/> <input type="button" value="Copy"/> <input type="button" value="Add New"/>		

Figure 16: Route Pattern for Extension Dialing

Cloud Connector Edition (10.64.4.149)

Cloud Connector first setup requires providing parameters such as SIP domains, IP addresses of virtual machines, configuration of REFER, and some other parameters in the INI file. The INI file is the main source of configuration information for Cloud Connector. During the install process, Cloud Connector installs the fresh set of Virtual machines and applies configuration settings from the INI file. Any parameters that are not present in INI are set to the default values.

Below is the snippet from INI file for Gateway Configuration.

```
;Parameters for gateway
;If only one Gateway is needed, remove entire [GateWay2] section. Don't keep it but leave
values empty.
;If Gateway FQDN uses 0365 Sip Domain in name for TLS purposes, be sure to set
InternetDNSIPAddress to allow Edge to resolve these records
[Gateway1]
; Gateway FQDN
FQDN=oracle.tekvizionlabs.local
;Gateway IP address
IP=10.64.4.149
;Gateway Port
Port=5061
;Protocol for SIP traffic (TCP or TLS)
Protocol=TLS
;List of voice routes used by this gateway.
;Routes are defined in the next section.
VoiceRoutes=LocalRoute

..... Parameters for hybrid voice routing .....
[HybridVoiceRoutes]
;Named voice route to be used by one or more gateways
LocalRoute=.*

..... Parameters for TrunkConfiguration .....
[TrunkConfiguration]
;Whether Gateways support Refer. It is used for Call Transfer scenario.
;The value can be "true" or "false". Default value is "true".
;EnableReferSupport set to "true" means the Gateway(s) support Refer which can handle all
the call transfer stuffs.
;EnableReferSupport set to "false" means the Gateway(s) don't support Refer. Then Mediation
Server will handle all the call transfer stuffs.
EnableReferSupport=true

;Whether forward PAI from Mediation Server to Gateways
;The value can be "true" or "false". Default value is "true".
ForwardPAI=false
```

Test Plan & Results

Test Plan

The testing was done with varied permutations/combinations and was performed by tekVizion.

The test plan consisted of the following test cases. All tests passed.

External ID	Title	Status	Comments
Inbound / Outbound / Extension Dialing			
1	Avaya 7.0 calls Cisco UCM 11.5 via Extension	Pass	
2	Avaya 7.0 calls Cisco UCM 11.5 using 10 digit calling	Pass	
3	Avaya 7.0 calls Microsoft Lync 2013 via Extension	Pass	
4	Avaya 7.0 calls Microsoft Lync 2013 using 10 digit calling	Pass	
5	Avaya 7.0 calls Skype for Business Online using 10 digit calling	Pass	Early media is disabled at ESBC to fix the RBT issue.
6	Cisco UCM 11.5 calls Avaya 7.0 via Extension	Pass	
7	Cisco UCM 11.5 calls Avaya 7.0 using 10 digit calling	Pass	
8	Cisco UCM 11.5 calls Microsoft Lync 2013 via Extension	Pass	
9	Cisco UCM 11.5 calls Microsoft Lync 2013 using 10 digit calling	Pass	
10	Cisco UCM 11.5 calls Skype for Business Online using 10 digit calling	Pass	
11	Microsoft Lync 2013 calls Avaya 7.0 via Extension	Pass	
12	Microsoft Lync 2013 calls Avaya 7.0 using 10 digit calling	Pass	
13	Microsoft Lync 2013 calls Cisco UCM 11.5 via Extension	Pass	
14	Microsoft Lync 2013 calls Cisco UCM 11.5 using 10 digit calling	Pass	
15	Microsoft Lync 2013 calls Skype for Business Online using 10 digit calling	Pass	
16	Skype for Business Online calls Avaya 7.0 via Extension	Pass	

17	Skype for Business Online calls Avaya 7.0 using 10 digit calling	Pass	
18	Skype for Business Online calls Cisco UCM 11.5 via Extension	Pass	
19	Skype for Business Online calls Cisco UCM 11.5 using 10 digit calling	Pass	
20	Skype for Business Online calls Microsoft Lync 2013 via Extension	Pass	
21	Skype for Business Online calls Microsoft Lync 2013 using 10 digit calling	Pass	
Transfer Functionality			
Each Transfer test case is executed in the following ways:			
Semi-Attended & Consultative transfer, based on re-INVITE & REFER methods			
22	PSTN calls into Avaya 7.0 and transfers to Cisco 11.5 using 10 digit calling	Pass	
23	PSTN calls into Avaya 7.0 and transfers to Microsoft Lync 2013 using 10 digit calling	Pass	Executed transfer scenarios based on re-INVITE method.
24	PSTN calls into Avaya 7.0 and transfers to Cisco 11.5 via Extension	Pass	
25	PSTN calls into Avaya 7.0 and transfers to Microsoft Lync 2013 via Extension	Pass	Executed transfer scenarios based on re-INVITE method.
26	PSTN calls into Cisco UCM 11.5 and transfers to Avaya 7.0 using 10 digit calling	Pass	Executed transfer scenarios based on re-INVITE method.
27	PSTN calls into Avaya 7.0 and transfers to Skype for Business Online using 10 digit calling	Pass	Early media is disabled at ESBC to fix the RBT issue.
28	PSTN calls into Cisco UCM 11.5 and transfers to Avaya 7.0 via Extension	Pass	Same as TC 27
29	PSTN calls into Cisco UCM 11.5 and transfers to Microsoft Lync 2013 using 10 digit calling	Pass	Same as TC 27
30	PSTN calls into Cisco UCM 11.5 and transfers to Microsoft Lync 2013 via Extension	Pass	Same as TC 27

31	PSTN calls into Cisco UCM 11.5 and transfers to Skype for Business Online using 10 digit calling	Pass	Same as TC 27
32	PSTN calls into Cisco UCM 11.5 and transfers to Skype for Business Online via Extension	N/A	Inbound calls to SFB Online user with Extension is Not Applicable since SFB Online user cannot have both Tel URI and Extension configured at the same time.
33	PSTN calls into Microsoft Lync 2013 and transfers to Avaya 7.0 using 10 digit calling	Conditional Pass	Conditional Pass. Transfer functionality works fine. However, when Lync user initiates Semi-attended transfer based on REFER method, call originator (PSTN User) does not hear the Ring back Tone when the transfer is completed.*****
34	PSTN calls into Microsoft Lync 2013 and transfers to Avaya 7.0 via Extension	Conditional Pass	Same as TC 33
35	PSTN calls into Microsoft Lync 2013 and transfers to Cisco 11.5 using 10 digit calling	Conditional Pass	Same as TC 33
36	PSTN calls into Microsoft Lync 2013 and transfers to Skype for Business Online using 10 digit calling	Pass	Executed transfer scenarios based on re-INVITE method.
37	PSTN calls into Microsoft Lync 2013 and transfers to Cisco 11.5 via Extension	Conditional Pass	Same as TC 33
Call Hold / Resume			
38	Avaya 7.0 calls Cisco 11.5 via Extension and places the call on hold & reconnects	Pass	
39	Avaya 7.0 calls Skype for Business Online via Extension and places the call on hold & reconnects	Pass	This test case is executed by dialing 10 Digits of the SFB Online user, since SFB Online cannot user have 'Tel URI and Extension Configured at the same time'.
40	Avaya 7.0 calls Microsoft Lync 2013 via Extension and places the call on hold & reconnects	Pass	
41	Cisco UCM 11.5 calls Avaya 7.0 via Extension and places the call on hold & reconnects	Pass	
42	Cisco UCM 11.5 calls Microsoft Lync 2013 via Extension and places the call on hold & reconnects	Pass	
43	Cisco UCM 11.5 calls Skype for Business Online via Extension and places the call on hold & reconnects	Pass	This test case is executed by dialing 10 Digits of the SFB Online user, since SFB Online user cannot have 'Tel URI and Extension Configured at the same time'.

44	Microsoft Lync 2013 calls Avaya 7.0 via Extension and places the call on hold & reconnects	Pass	
45	Microsoft Lync 2013 calls Cisco UCM 11.5 via Extension and places the call on hold & reconnects	Pass	
46	Microsoft Lync 2013 calls Skype for Business Online via Extension and places the call on hold & reconnects	Pass	
47	Skype for Business Online calls Avaya 7.0 via Extension and places the call on hold & reconnects	Pass	
48	Skype for Business Online calls Cisco UCM 11.5 via Extension and places the call on hold & reconnects	Pass	
Serial Forking w/ LDAP Integration			
49	Avaya 7.0 calls a user (via 10 digit dial) which is configured on both Lync 2013 and SFB. Both Lync 2013 and SFB instances should ring in serially and either can be answered.	Pass	
50	Avaya 7.0 calls a user (via extension dial) which is configured on both Lync 2013 and SFB. Both Lync 2013 and SFB instances should ring in serially and either can be answered.	Pass	Extension dial for SFB Online user is NA, hence this test case is executed by converting Extension to 10 Digit DID for SFB Online user at E-SBC.
51	Cisco UCM 11.5 calls a user (via 10 digit dial) which is configured on both Lync 2013 and SFB. Both Lync 2013 and SFB instances should ring in serially and either can be answered.	Pass	
52	Cisco UCM 11.5 calls a user (via extension dial) which is configured on both Lync 2013 and SFB. Both Lync 2013 5 and SFB instances should ring in serially and either can be answered.	Pass	Extension dial for SFB Online user is NA, hence this test case is executed by converting Extension to 10 Digit DID for SFB Online user at E-SBC.
CCE Specific Test Cases			

53	Call forward always from online user to PSTN	Pass	Executed this test case by originating the call from Cisco UCM
54	Call forward always from online user to early media number	Pass	
55	Call Forward no answer from Online user to PSTN	Pass	
56	Call Forward no answer from Online user to Voicemail	Pass	
57	Simultaneous ring setup with Online user with another Online user	Pass	
58	Simultaneous ring setup on Online with PSTN	Pass	
59	Call from CCE user Early Media IVR number and test DTMF before 200 OK	Pass	
60	Call from CCE user to toll free number and test DTMF	Pass	
61	Anonymous call from Online user to PSTN	Pass	

Appendix A

Full SBC configuration

```
show running-config short
certificate-record
    name                LyncMed
    state               TX
    locality            Plano
    common-name         medserver1.lyncclabsram.local
    digest-algor       sha1
certificate-record
    name                LyncRoot
    state               TX
    locality            Plano
    common-name         lyncclabsram-DC-CA
    digest-algor       sha1
certificate-record
    name                SBCCCE2
    state               TX
    locality            Plano
    common-name         oracle.tekvizionlabs.local
    digest-algor       sha1
certificate-record
    name                SBCCert
    state               TX
    locality            Plano
    common-name         oracle.tekvizionlabs.local
    digest-algor       sha1
certificate-record
    name                SBCCertCCE
    state               TX
    locality            Plano
    common-name         oracle.tekvizionlabs.local
    digest-algor       sha1
certificate-record
    name                cceRoot
    state               TX
    locality            Plano
    common-name         SfB CCE Root
    digest-algor       sha1
certificate-record
    name                newCCERoot
    state               TX
    locality            Plano
    common-name         SfB CCE Root
    digest-algor       sha1
certificate-record
    name                rootcert
    state               TX
    locality            Plano
    common-name         lyncclabsram-DC-CA
    digest-algor       sha1
codec-policy
    name                G711Only
    allow-codecs        * G729:no
```

```

    add-codecs-on-egress          PCMU PCMA telephone-event
codec-policy
    name                          G729only
    allow-codecs                  * PCMA:no PCMU:no
    add-codecs-on-egress          G729
codec-policy
    name                          Xcode
    allow-codecs                  *
    add-codecs-on-egress          CN
codec-policy
    name                          toSFB
    allow-codecs                  *
    add-codecs-on-egress          CN
filter-config
    name                          all
host-route
    dest-network                  155.212.214.199
    netmask                       255.255.255.0
    gateway                       192.65.79.97
ldap-config
    name                          ldapCfg
    ldap-servers                  172.16.31.91:389
    realm                        inside
    username                      administrator@lynclabsram.local
    password                      *****
    ldap-search-base              CN=Users,DC=lynclabsram,DC=local
    ldap-transactions
        route-mode                attribute-order-only
        ldap-cfg-attributes
            name                    msRTCSIP-Line
            next-hop                med1.lynclabsram.local
            realm                    inside
            extraction-regex        (.*)
            value-format            tel:$0
        ldap-cfg-attributes
            name                    telephoneNumber
            next-hop
mediationserver.tekvizionlabs.local
    realm                        outside
    extraction-regex              (.*)
    value-format                  tel:$0
local-policy
    from-address                  *
    to-address                    25
                                52600025
    source-realm                  *
    policy-attribute
        next-hop                  10.89.17.7
        realm                      AvayaRealm
local-policy
    from-address                  *
    to-address                    26
                                52700026
    source-realm                  *
    policy-attribute
        next-hop                  10.80.18.3

```



```

realm CUCMRealm
local-policy
  from-address *
  to-address 27
  source-realm 52800027
  policy-attribute
    next-hop med1.lyncclabsram.local
    realm inside
local-policy
  from-address *
  to-address 28
  source-realm 52900028
  policy-attribute
    next-hop mediationserver.tekvizionlabs.local
    realm outside
local-policy
  from-address *
  to-address *
  source-realm AvayaRealm
  state CUCMRealm
  policy-attribute
    next-hop inside
    realm disabled
local-policy
  from-address *
  to-address 2142425
  source-realm 800
  description 9725
  policy-attribute
    next-hop topstn
    realm 10.64.1.72
local-policy
  from-address *
  to-address 2432
  source-realm 2601
  policy-attribute
    next-hop 5270002601
    realm ldap:ldapCfg
local-policy
  from-address *
  to-address med1.lyncclabsram.local
  source-realm *
  policy-attribute
    next-hop med1.lyncclabsram.local
    realm inside
local-policy
  from-address *
  to-address mediationserver.tekvizionlabs.local

```

```

source-realm *
policy-attribute
  next-hop mediationserver.tekvizionlabs.local
  realm outside
local-policy
  from-address *
  to-address 9876543210
  source-realm *
  policy-attribute
    next-hop 172.16.29.71
    realm toSFB
local-policy
  from-address *
  to-address *
  source-realm outside
  policy-attribute
    next-hop med1.lyncclabsram.local
    realm inside
    methods OPTIONS
media-manager
  initial-guard-timer 86400
  options xcode-gratuitous-rtcp-report-generation
media-sec-policy
  name RTP
media-sec-policy
  name SRTP
  inbound
    profile SRTP
    mode srtp
    protocol sdes
  outbound
    profile SRTP
    mode srtp
    protocol sdes
network-interface
  name s0p0
  description WAN
  ip-address 10.64.4.149
  netmask 255.255.0.0
  gateway 10.64.1.1
  gw-heartbeat
    state enabled
  dns-ip-primary 10.64.4.61
  dns-domain tekvizionlabs.local
  hip-ip-list 10.64.4.149
  10.64.4.145
  10.64.4.147
  icmp-address 10.64.4.149
  10.64.4.145
  10.64.4.147
network-interface
  name s0p1
  description LAN
  ip-address 10.70.50.20
  netmask 255.255.255.0
  gateway 10.70.50.1

```

```

dns-ip-primary      172.16.31.91
dns-domain          lynclabstram.local
hip-ip-list         10.70.50.20
                   10.70.50.21
                   10.70.50.22
                   10.70.50.23
icmp-address        10.70.50.20
                   10.70.50.21
                   10.70.50.22
                   10.70.50.23
phy-interface
  name              s0p0
  operation-type    Media
phy-interface
  name              s0p1
  operation-type    Media
  port              1
playback-config
  name              transferrbt
  entry
    encoding        PCMU
    filename         US_ringbackPCMU.raw
realm-config
  identifier        AvayaRealm
  network-interfaces s0p1:0
  media-sec-policy RTP
  early-media-allow none
  codec-policy      Xcode
  rtcp-policy       rtcpGen
realm-config
  identifier        CUCMRealm
  network-interfaces s0p1:0
  media-sec-policy RTP
  codec-policy      G729only
  codec-manip-in-realm enabled
  rtcp-policy       rtcpGen
realm-config
  identifier        PSTN
  network-interfaces s0p0:0
  media-sec-policy RTP
  codec-policy      G711Only
  rtcp-policy       rtcpGen
realm-config
  identifier        inside
  network-interfaces s0p1:0
  media-sec-policy RTP
  spl-options        comfort-noise-generate
  codec-policy      G711Only
  rtcp-policy       rtcpGen
realm-config
  identifier        outside
  network-interfaces s0p0:0
  media-sec-policy S RTP
  spl-options        comfort-noise-generate
  codec-policy      G711Only
  rtcp-policy       rtcpGen

```

```

realm-config
  identifier                               outside-rtp
  network-interfaces                       s0p0:0
realm-config
  identifier                               toSFB
  network-interfaces                       s0p1:0
  media-sec-policy                         RTP
  codec-policy                             Xcode
  rtcp-policy                              rtcpGen
response-map
  name                                     change183to180
  entries
    rcv-code                               183
    xmit-code                              180
    reason                                 Ringing
rtcp-policy
  name                                     rtcpGen
  rtcp-generate                            all-calls
sdes-profile
  name                                     SRTP
  use-ingress-session-params              srtcp-encrypt
session-agent
  hostname                                 10.64.1.72
  ip-address                               10.64.1.72
  transport-method                         StaticTCP
  realm-id                                 PSTN
  description                              PSTN
  ping-method                              OPTIONS;hops=0
  ping-interval                            30
session-agent
  hostname                                 10.80.18.3
  ip-address                               10.80.18.3
  transport-method                         StaticTCP
  realm-id                                 CUCMRealm
  description                              CUCM
  ping-method                              OPTIONS;hops=0
  ping-interval                            30
session-agent
  hostname                                 10.89.17.7
  ip-address                               10.89.17.7
  transport-method                         StaticTCP
  realm-id                                 AvayaRealm
  description                              Avaya
  ping-method                              OPTIONS;hops=0
  ping-interval                            30
session-agent
  hostname                                 172.16.29.71
  ip-address                               172.16.29.71
  transport-method                         StaticTCP
  realm-id                                 toSFB
  ping-method                              OPTIONS
  ping-interval                            30
session-agent
  hostname                                 med1.lyncclabsram.local
  ip-address                               172.16.31.97
  port                                     5067

```

```

transport-method      StaticTLS
realm-id              inside
ping-method           OPTIONS;hops=0
ping-interval         30
refer-call-transfer   enabled
session-agent
  hostname             mediationserver.tekvizionlabs.local
  ip-address           10.64.4.59
  port                 5067
  transport-method     StaticTLS
  realm-id              outside
  ping-method           OPTIONS;hops=0
  ping-interval        30
  in-translationid     delplus1
  refer-call-transfer   enabled
session-translation
  id                   addPlus
  rules-calling         addPlus1
  rules-called          addPlus1
session-translation
  id                   delplus1
  rules-calling         delPlus1
  rules-called          delPlus1
sip-config
  home-realm-id        inside
  registrar-domain     *
  registrar-host       *
  registrar-port       5060
  options              max-udp-length=0
sip-interface
  realm-id             AvayaRealm
  sip-port
    address             10.70.50.21
    transport-protocol  TCP
    allow-anonymous     agents-only
  out-manipulationid  NAT_IP_Avaya
  response-map         change183to180
sip-interface
  realm-id             CUCMRealm
  sip-port
    address             10.70.50.22
    transport-protocol  TCP
    allow-anonymous     agents-only
  secured-network      enabled
  out-manipulationid  NAT_IP
sip-interface
  realm-id             PSTN
  sip-port
    address             10.64.4.145
    transport-protocol  TCP
    allow-anonymous     agents-only
  spl-options          playback-on-refer="transferrbt"
  out-manipulationid  NAT_IP
sip-interface
  realm-id             inside
  sip-port

```

```

        address 10.70.50.20
        transport-protocol TCP
        allow-anonymous agents-only
    sip-port
        address 10.70.50.20
        port 5061
        transport-protocol TLS
        tls-profile LyncTLS
        allow-anonymous agents-only
    secured-network enabled
    spl-options playback-on-refer="transferrbt"
    in-manipulationid Stripsdp183
    out-manipulationid NatIpLync
sip-interface
    realm-id outside
    sip-port
        address 10.64.4.149
        transport-protocol TCP
        allow-anonymous agents-only
    sip-port
        address 10.64.4.149
        port 5061
        transport-protocol TLS
        tls-profile CCETLS
        allow-anonymous agents-only
    in-manipulationid Stripsdp183
    out-manipulationid NatIpCce
sip-interface
    realm-id outside-rtp
    sip-port
        address 10.64.4.147
        transport-protocol TCP
sip-interface
    realm-id toSFB
    sip-port
        address 10.70.50.23
        transport-protocol TCP
        allow-anonymous agents-only
sip-manipulation
    name Changeinactosendonly
    description Change inactive to sendonly for pstn tran
    header-rule
        name changeSDP
        header-name Content-Type
        action manipulate
        msg-type request
        methods INVITE
        element-rule
            name inacttosendonly
            parameter-name application/sdp
            type mime
            action find-replace-all
            comparison-type pattern-rule
            match-value a=inactive
            new-value a=sendonly
sip-manipulation

```

```

name NAT_IP
header-rule
  name
  header-name From
  action From
  msg-type manipulate
  element-rule request
    name fromEr
    type uri-host
    action find-replace-all
    new-value oracle.tekvizionlabs.local
header-rule
  name To
  header-name To
  action manipulate
  msg-type request
  element-rule
    name ToEr
    type uri-host
    action find-replace-all
    new-value $REMOTE_IP
header-rule
  name RequestURI
  header-name request-uri
  action manipulate
  methods INVITE
  element-rule
    name ruriEr
    type uri-host
    action find-replace-all
    new-value $REMOTE_IP
header-rule
  name alterPAI
  header-name P-Asserted-Identity
  action manipulate
  element-rule
    name alterPAIEr
    type uri-host
    action find-replace-all
    new-value $LOCAL_IP
header-rule
  name msSource
  header-name ms-call-source
  action delete
  methods INVITE
sip-manipulation
  name NAT_IP_Avaya
  header-rule
    name From
    header-name From
    action manipulate
    msg-type request
    element-rule
      name fromEr
      type uri-host
      action find-replace-all

```

```

        new-value                oracle.tekvizionlabs.local
    element-rule
        name                      store_user
        type                      uri-user
        action                    store
header-rule
    name                          To
    header-name                   To
    action                        manipulate
    msg-type                      request
    element-rule
        name                      ToEr
        type                      uri-host
        action                    find-replace-all
        new-value                 $REMOTE_IP
header-rule
    name                          RequestURI
    header-name                   request-uri
    action                        manipulate
    methods                      INVITE
    element-rule
        name                      ruriEr
        type                      uri-host
        action                    find-replace-all
        new-value                 $REMOTE_IP
header-rule
    name                          delPAI
    header-name                   P-Asserted-Identity
    action                        delete
    msg-type                      request
    methods                      INVITE
header-rule
    name                          msSource
    header-name                   ms-call-source
    action                        delete
    methods                      INVITE
header-rule
    name                          createPAI
    header-name                   Contact
    action                        manipulate
    msg-type                      request
    methods                      INVITE
    match-value
    element-rule
        name                      updateContact
        type                      uri-user
        action                    find-replace-all
        new-value                 $From.$store_user.$0
header-rule
    name                          storeContact
    header-name                   Contact
    action                        store
    comparison-type              pattern-rule
    msg-type                      request
    methods                      INVITE
header-rule

```



```

        name addPAI
        header-name P-Asserted-Identity
        action add
        comparison-type boolean
        msg-type request
        methods INVITE
        match-value $delPAI
        new-value $storeContact.$0
sip-manipulation
    name NatIpCce
    header-rule
        name From
        header-name From
        action manipulate
        methods INVITE
        element-rule
            name fromEr
            type uri-host
            action find-replace-all
            new-value oracle.tekvizionlabs.local
    header-rule
        name To
        header-name To
        action manipulate
        methods INVITE
        element-rule
            name ToEr
            type uri-host
            action find-replace-all
            new-value
mediationserver.tekvizionlabs.local
    header-rule
        name RequestURI
        header-name request-uri
        action manipulate
        methods INVITE
        element-rule
            name ruriEr
            type uri-host
            action find-replace-all
            new-value
mediationserver.tekvizionlabs.local
    header-rule
        name alterPAI
        header-name P-Asserted-Identity
        action manipulate
        element-rule
            name alterPAIEr
            type uri-host
            action find-replace-all
            new-value $LOCAL_IP
    header-rule
        name Contact
        header-name Contact
        action manipulate
        methods INVITE

```

```

        element-rule
            name                contactEr
            type                 uri-host
            action               find-replace-all
            new-value            oracle.tekvizionlabs.local

header-rule
    name                AddPlus1
    header-name         referred-by
    action              manipulate
    element-rule
        name                addplus1
        type                 uri-user
        action               find-replace-all
        new-value            "+1"+$ORIGINAL

header-rule
    name                alterreferby
    header-name         From
    action              sip-manip
    msg-type            request
    new-value           alterreferby

header-rule
    name                removePC
    header-name         From
    action              manipulate
    methods             INVITE
    element-rule
        name                removePCEr
        parameter-name       phone-context
        type                 uri-user-param
        action               delete-element

header-rule
    name                from_user
    header-name         from
    action              manipulate
    methods             INVITE
    element-rule
        name                fromEr
        type                 uri-user
        action               find-replace-all
        match-value         ^(\d{10})
        new-value           "+1"+$ORIGINAL

header-rule
    name                Plus1RURI
    header-name         Request-URI
    action              manipulate
    element-rule
        name                PlusRURIEr
        type                 uri-user
        action               find-replace-all
        match-value         ^(\d{10})
        new-value           "+1"+$ORIGINAL

        element-rule
            name                PlusRURI_store
            type                 uri-user
            action               store

header-rule

```

```

        name                               Plus1To
        header-name                         To
        action                              manipulate
        element-rule
            name                             PlusToEr
            type                             uri-user
            action                           find-replace-all
            match-value                      ^(\d{10})
            new-value                        "+1"+$ORIGINAL
        element-rule
            name                             copyRURI
            type                             uri-user
            action                           find-replace-all
            match-value                      ^(\d{4})
            new-value
$Plus1RURI.$PlusRURI_store.$0
sip-manipulation
    name                                   NatIpLync
    header-rule
        name                               from
        header-name                         From
        action                              sip-manip
        new-value                          NAT_IP
    header-rule
        name                               alterPAI
        header-name                         P-Asserted-Identity
        action                              manipulate
        methods                             INVITE
        element-rule
            name                             alterPAIEr
            type                             uri-host
            action                           find-replace-all
            new-value                        oracle.tekvizionlabs.local
    header-rule
        name                               alterContact
        header-name                         Contact
        action                              manipulate
        methods                             INVITE
        element-rule
            name                             alterContactEr
            type                             uri-host
            action                           find-replace-all
            new-value                        oracle.tekvizionlabs.local
    header-rule
        name                               To
        header-name                         To
        action                              manipulate
        methods                             INVITE
        element-rule
            name                             toEr
            type                             uri-host
            action                           find-replace-all
            new-value                        med1.lyncclabsram.local
    header-rule
        name                               ruri
        header-name                         request-uri

```

```

        action                manipulate
        methods                INVITE
        element-rule
            name                ruriEr
            type                uri-host
            action              find-replace-all
            new-value           med1.lyncclabsram.local
header-rule
    name                      alterReferredBy
    header-name               Referred-By
    action                    manipulate
    msg-type                  request
    methods                   INVITE
    element-rule
        name                  alterRB
        type                  uri-host
        action                find-replace-all
        new-value             oracle.tekvizionlabs.local
sip-manipulation
    name                      Stripsdp183
    description                For incoming 183 from Lync, strip SDP
    header-rule
        name                  check183
        header-name           @status-line
        action                 store
        comparison-type       pattern-rule
        element-rule
            name              is183
            type              status-code
            action            store
            comparison-type   pattern-rule
            match-value       183
        header-rule
            name              delSDP
            header-name       Content-Type
            action            manipulate
            comparison-type   case-insensitive
            match-value       $check183.$is183
            element-rule
                name          del183SDP
                parameter-name application/sdp
                type          mime
                action        delete-element
                comparison-type boolean
        header-rule
            name              delContentType
            header-name       Content-Type
            action            manipulate
            comparison-type   boolean
            match-value       $check183.$is183
            element-rule
                name          delCT
                parameter-name *
                type          header-param
                action        delete-header
header-rule

```

name	inactosendonly
header-name	From
action	sip-manip
msg-type	request
new-value	Changeinactosendonly
sip-manipulation	
name	alter4xxOPTIONS
header-rule	
name	alterRURI
header-name	Request-URI
action	manipulate
msg-type	reply
methods	OPTIONS
element-rule	
name	alterRURI_statusCode
type	status-code
action	find-replace-all
match-value	483 404
new-value	200
element-rule	
name	alterRURI_ReasonPhrase
type	reason-phrase
action	find-replace-all
match-value	Too Many Hops Not Found
new-value	OK
sip-manipulation	
name	delReqNatIp
header-rule	
name	delRequire
header-name	Require
action	delete
msg-type	reply
methods	INVITE
header-rule	
name	nat
header-name	From
action	sip-manip
new-value	NAT_IP
sip-manipulation	
name	alterreferby
header-rule	
name	checkReferBy
header-name	REFERRED-BY
action	manipulate
element-rule	
name	checkReferBy
type	uri-host
action	find-replace-all
new-value	oracle.tekvizionlabs.local
sip-manipulation	
name	stripRR
join-headers	Record-Route
header-rule	
name	delRR
header-name	Record-Route
action	delete

```

sip-monitoring
  match-any-filter          enabled
  monitoring-filters        all
steering-pool
  ip-address                10.64.4.145
  start-port                49600
  end-port                  65535
  realm-id                  PSTN
steering-pool
  ip-address                10.64.4.147
  start-port                49600
  end-port                  65535
  realm-id                  outside-rtp
steering-pool
  ip-address                10.64.4.149
  start-port                49600
  end-port                  65535
  realm-id                  outside
steering-pool
  ip-address                10.70.50.20
  start-port                49562
  end-port                  65535
  realm-id                  inside
steering-pool
  ip-address                10.70.50.21
  start-port                49600
  end-port                  65535
  realm-id                  AvayaRealm
steering-pool
  ip-address                10.70.50.22
  start-port                49600
  end-port                  65535
  realm-id                  CUCMRealm
steering-pool
  ip-address                10.70.50.23
  start-port                49562
  end-port                  65535
  realm-id                  toSFB
system-config
  process-log-level         DEBUG
  default-gateway           192.65.79.97
tls-global
  session-caching           enabled
tls-profile
  name                      CCETLS
  end-entity-certificate    SBCCCE2
  trusted-ca-certificates   newCCERoot
  mutual-authenticate       enabled
  tls-version                tlsv12
tls-profile
  name                      LyncTLS
  end-entity-certificate    SBCCert
  trusted-ca-certificates   rootcert
                           LyncRoot
  mutual-authenticate       enabled
  tls-version                tlsv1

```

```
translation-rules
  id          addPlus1
  type        add
  add-string  +1
translation-rules
  id          delPlus1
  type        delete
  delete-string +1
web-server-config
office365#
```

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/SFB Server, and the Oracle ECB and 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.

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/SFB Server mediation server, the Lync/SFB 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 and/or SFB 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/SFB Server Front End server;
- A Lync/SFB Server Standard Edition Server; and
- A Lync/SFB Server Mediation Server.

On the Oracle ECB and E-SBC

The Oracle SBC and ECB provide a rich set of statistical counters available from the CLI, 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 console:

```
oraclesbc1# reset sipd
oraclesbc1# notify sipd debug
oraclesbc1#
enabled SIP Debugging
oraclesbc1# notify all rotate-logs
```

Examining the log files

Note: You will FTP to the management interface of the ECB or 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 E-SBC and ECB, and provide traces in a common log format for local viewing or for exporting to your PC. Please check the “Monitor and Trace SIP Messages” section (page 140) of the E-SBC Web GUI User Guide available at http://docs.oracle.com/cd/E56581_01/index.htm. For the ECB, see the “Monitor and Trace” section (page 95) of the User’s Guide available at http://docs.oracle.com/cd/E55725_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.

All devices tested in this document will listen on TCP port 5060 for SIP signaling. In our example we are listening on 5060 on the PSTN facing NIC. Tests may include:

- Client to pool server: `telnet <servername> 5060`
- Pool server to Mediation Server: `telnet <servername> 5060`

Cisco Real-Time Monitoring Tool (RTMT)

The Cisco Real-Time Monitoring Tool (RTMT) is a tool that can be downloaded from CUCM to a Windows or Linux computer. See <https://supportforums.cisco.com/document/93281/using-rtmt-monitor-cisco-unity-connection-and-cucm> for details.

Appendix B

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.

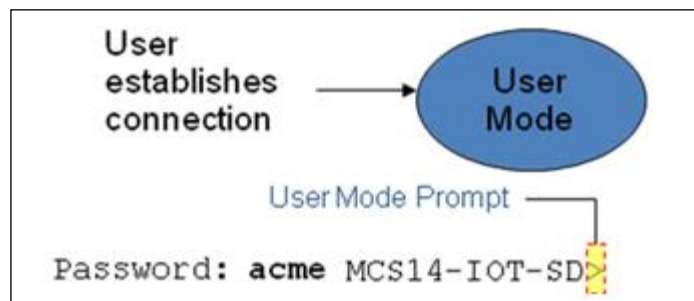
Initial connectivity will be through the serial console port. At a minimum, this is how to configure the management (eth0) interface on the SBC.

ACLI Basics

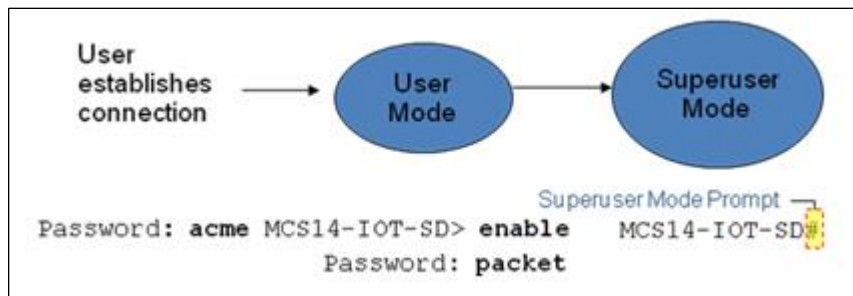
There are two password protected modes of operation within the ACLI, User mode and Superuser mode.

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.



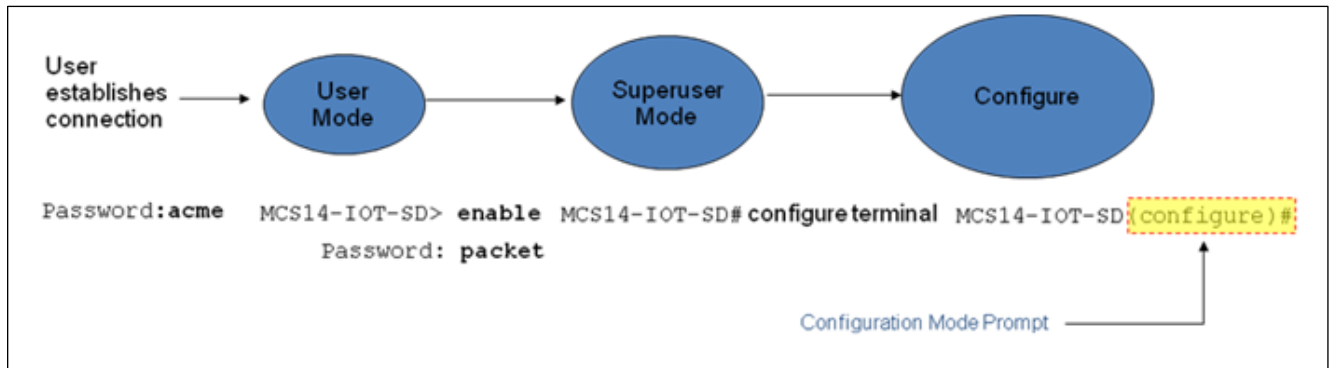
The Superuser mode allows for access to all system commands for operation, maintenance, and administration. This mode is identified by the pound sign (#) in the prompt after the target name. To enter the Superuser mode, issue the enable command in the User mode.



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.

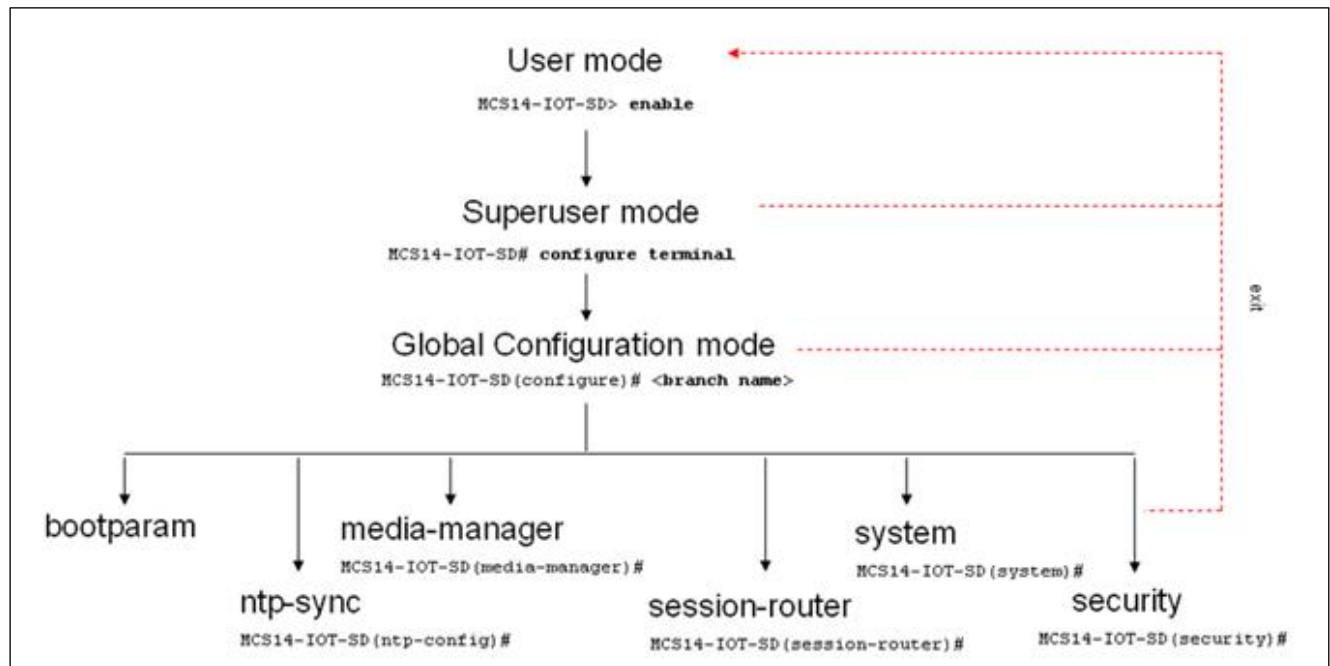
You must enter the Configuration mode to configure elements. For example, you can access the configuration branches and configuration elements for signaling and media configurations. To enter the Configuration mode, issue the `configure terminal` command in the Superuser mode.

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.



The ntp-sync and bootparams branches are flat branches (i.e., they do not have elements inside the branches). The rest of the branches have several elements under each of the branches.

The bootparam branch provides access to SBC boot parameters.

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, ivf-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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