



## Oracle Enterprise Session Border Controller and CUCM 11.0 with Telus Enterprise SIP Trunk

Technical Application Note



## Disclaimer

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## **Intended Audience**

This document is intended for use by Oracle personnel, third party Systems Integrators, and end users of the Oracle Enterprise Session Border Controller (E-SBC). It assumes that the reader is familiar with basic operations of the Oracle Enterprise Session Border Controller – Acme Packet 4600 / Acme Packet 3820.

## **Document Overview**

Cisco Call Manager offers the ability to connect to Internet telephony service providers (ITSP) using an IP-based SIP trunk. This reduces the cost and complexity of extending an enterprise's telephony system outside its network borders. Oracle Enterprise Session Border Controllers (E-SBCs) play an important role in SIP trunking as they are used by many ITSPs and some enterprises as part of their SIP trunking infrastructure.

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

## Introduction

### Audience

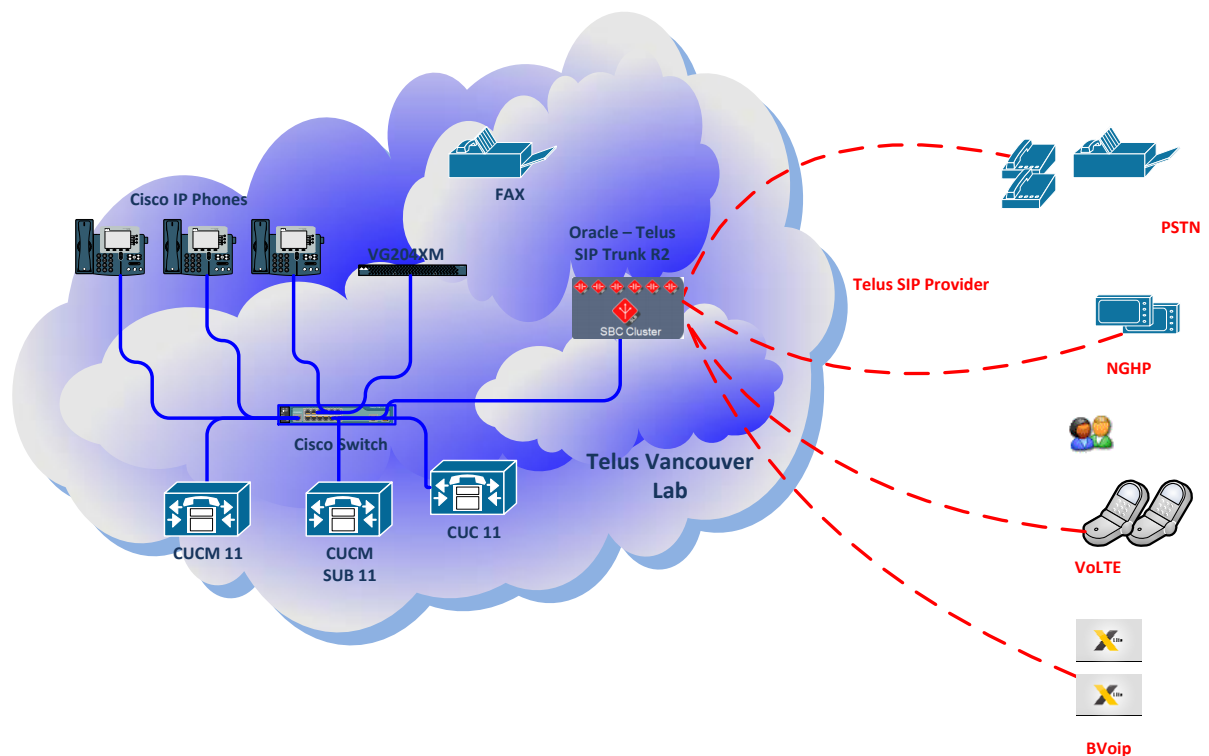
This is a technical document intended for telecommunications engineers with the purpose of configuring the Oracle Enterprise Session Border Controller and CUCM. There will be steps that require navigating the Command Line Interface (CLI). Understanding the basic concepts of TCP/UDP, IP/Routing, SIP/RTP, TLS and SRTP are also necessary to complete the configuration and for troubleshooting, if necessary.

### Requirements

- Fully installed and configured Cisco Call Manager 11.0
- Oracle Enterprise Session Border Controller is running ECZ7.3.0 m1. Note: the configuration running on the E-SBC is backward/forward compatible with any release in the 7.3.0 stream.
- Telus IP trunk based customers with dedicated data connectivity to Telus.

### Architecture

- The following reference architecture shows a logical view of the connectivity between CUCM, E-SBC and the Telus trunk.



## Configuring the Oracle **Enterprise** Session Border Controller

In this section we describe the steps for configuring an Oracle Enterprise Session Border Controller, formally known as an Acme Packet Net-Net Enterprise Session Director, for use with CUCM Server in a SIP trunking scenario.

### **In Scope**

The following guide configuring the Oracle E-SBC assumes that this is a newly deployed device dedicated to a single customer. If a service provider currently has the E-SBC deployed then please refer to the ACLI Configuration Guide on [http://docs.oracle.com/cd/E56581\\_01/index.htm](http://docs.oracle.com/cd/E56581_01/index.htm) for a better understanding of the Command Line Interface (CLI).

Note that Oracle offers several models of E-SBC. This document covers the setup for the E-SBC platform running ECZ7.3.0 or later. If instructions are needed for other Oracle E-SBC models, please contact your Oracle representative.

### **Out of Scope**

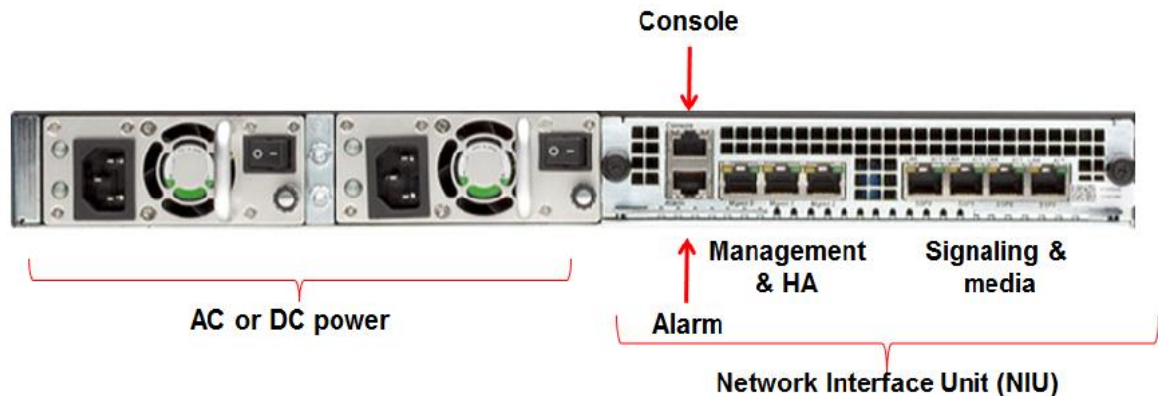
- Configuration of Network management including SNMP and RADIUS

### **What will you need**

- Hypervisor with console connectivity through the hypervisor
- Terminal emulation application such as PuTTY or HyperTerm
- Passwords for the User and Super user modes on the Oracle E-SBC
- IP address to be assigned to management interface (Wancom0) of the E-SBC - the Wancom0 management interface must be connected and configured to a management network separate from the service interfaces. Otherwise the E-SBC is subject to ARP overlap issues, loss of system access when the network is down, and compromising DDoS protection. Oracle does not support E-SBC configurations with management and media/service interfaces on the same subnet.
- IP address of CUCM external facing NIC
- IP addresses to be used for the E-SBC internal and external facing ports (Service Interfaces)
- IP address of the next hop gateway in the service provider network

## Configuring the E-SBC

Once the Oracle SBC is racked and the power cable connected, you are ready to set up physical network connectivity.



Plug the slot 0 port 0 (s0p0) interface into your outside (Telus next-hop facing) network and the slot 1 port 0 (s1p0) interface into your inside (CUCM server-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 `SBC1` are parameters which are specific to an individual deployment. Note: The ACLI is case sensitive.

### Establish the serial connection and logging in the SBC

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

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

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

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

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

```
Password: acme
SBC1> enable
Password: packet
SBC1# configure terminal
SBC1 (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 E-SBC by going to SBC1#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

```
SBC1#(configure)bootparam
'.' = clear field; '-' = go to previous field; q = quit
boot device          : eth0
processor number     : 0
host name            : acmesystem
file name            : /boot/nnECZ730mlp1.32.bz --- >location where the
software is loaded on the SBC
inet on ethernet (e) : 192.168.1.22:ffffff80 --- > This is the ip
address of the management interface of the SBC, type the IP address and
mask in hex
inet on backplane (b) :
host inet (h)         :
gateway inet (g)      : 192.168.1.1 -> gateway address here
user (u)              : vxftp
ftp password (pw) (blank = use rsh) : vxftp
flags (f)             :
target name (tn)      : SBC1 -> ACLI prompt name & HA peer name
startup script (s)    :
other (o)             :
```

### Configuring the E-SBC

The following section walks you through configuring the Oracle E-SBC. It is outside the scope of this document to include all of the configuration elements as it will differ in every deployment.

### Header manipulation rule required for Telus

The header manipulation rule towardstrunk deletes the Require:100rel header from all SIP messaging which is going towards the Telus trunk. The sip-feature 100rel is added to the config for SIP PRACK interworking, and 100rel-interworking is added as OPTIONS on the trunk and core side sip-interface.



sip-manipulation	
name	towardstrunk
header-rule	
name	DelReq
header-name	Require
action	delete
match-value	100rel
sip-feature	
name	100rel
realm	Peer
support-mode-inbound	Pass
require-mode-inbound	Pass
proxy-require-mode-inbound	Pass
support-mode-outbound	Pass
require-mode-outbound	Pass
proxy-require-mode-outbound	Pass
last-modified-by	admin@192.168.20.101
last-modified-date	2016-05-05 17:18:13

## Webserver Configuration

A webserver is available on all Enterprise versions of Oracle E-SBCs. The Webserver can be used to provide tracing, configuration and dashboard info. For tracing info, 2 parts must be configured.

- The webserver must be enabled.
- Tracing filters must be applied.

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

sip-monitoring	
match-any-filter	disabled
state	enabled
short-session-duration	0
monitoring-filters	*
trigger-window	30

## Test Plan

### PSTN test cases

Test Number	Test Details	Pass/Fail/NA - Not Applicable
<b>Test with PSTN line</b>		
<b>Basic inbound/outbound call</b>		
TELUS_TC1.1	Call from PSTN phone to IP PBX phone 1. Confirm 2-way voice 2. Confirm the proper calling number is shown 3. Confirm the proper call display name is shown	Pass
TELUS_TC1.2	Call from IP PBX phone to PSTN phone 1. Confirm 2-way voice 2. Confirm the proper calling number is shown 3. Confirm the proper call display name is shown	Pass
<b>Basic inbound/outbound call with privacy</b>		
TELUS_TC1.4	Call from IP PBX phone to PSTN phone, when dialling from the IP PBX phone, use the prefix if applicable to temporary suppress the call display 1. Confirm 2-way voice 2. Confirm the proper calling number is not shown 3. Confirm the proper call display name is not shown	Pass
<b>Hold and resume</b>		
TELUS_TC1.5	Call from PSTN to IP PBX - after the call setup the PBX phone puts the call on-hold or (MOH), waits 30 seconds, resumes. Confirm audio both way after resume.	Pass
TELUS_TC1.6	Call from IP PBX to PSTN - after the call setup, use PSTN phone to put the call on-hold, wait 30 seconds, resume. Confirm audio both way after resume.	Pass
<b>Call Transfer (Blind transfer)</b>		
TELUS_TC1.7	IP PBX phone 1 calls IP PBX phone 2 IP PBX phone 2 performs a blind transfer to PSTN phone Confirm audio both way after the transfer	Pass
TELUS_TC1.9	PSTN phone calls IP PBX phone 1 IP PBX phone 1 performs a blind transfer to IP PBX phone 2 Confirm audio both way after the transfer	Pass
TELUS_TC1.11	PSTN phone calls IP PBX phone 1 IP PBX phone 1 performs a blind transfer to another PSTN Confirm both way audio.	Pass
<b>Call Transfer (Consult transfer)</b>		
TELUS_TC1.13	IP PBX phone 1 calls IP PBX phone 2 IP PBX phone 2 performs a consult transfer to PSTN phone Confirm audio both way after the transfer	Pass

TELUS_TC1.15	PSTN phone calls IP PBX phone 1 IP PBX phone 1 performs a consult transfer to IP PBX phone 2 Confirm audio both way after the transfer	Pass
TELUS_TC1.17	PSTN phone calls IP PBX phone 1 IP PBX phone 1 performs a consult transfer to another PSTN Confirm both way audio.	Pass
<b>Call Forwarding Unconditional</b>		
TELUS_TC1.19	Configure IP PBX phone 1 to CFU to PSTN phone IP PBX phone 2 calls phone 1 and should CFU to PSTN phone 1. Confirm 2-way voice 2. Confirm phone 1 number and display at PSTN phone	Pass
TELUS_TC1.20	Configure IP PBX phone 1 to CFU to PSTN phone from PSTN calls phone 1 and should CFU to PSTN phone 1. Confirm 2-way voice 2. Confirm phone 1 number and display at PSTN phone	Pass
<b>Call Forwarding Busy</b>		
TELUS_TC1.21	Configure IP PBX phone 1 to CFB to PSTN phone IP PBX phone 2 calls phone 1 and should CFB to PSTN phone 1. Confirm 2-way voice 2. Confirm phone 1 number and display at PSTN phone	Pass
TELUS_TC1.22	Configure IP PBX phone 1 to CFB to PSTN phone from PSTN calls phone 1 and should CFB to PSTN phone 1. Confirm 2-way voice 2. Confirm phone 1 number and display at PSTN phone	Pass
<b>Call Forwarding Don't Answer</b>		
TELUS_TC1.23	Configure IP PBX phone 1 to CFDA to PSTN phone IP PBX phone 2 calls phone 1 and should CFDA to PSTN phone 1. Confirm 2-way voice 2. Confirm phone 1 number and display at PSTN phone	Pass
TELUS_TC1.24	Configure IP PBX phone 1 to CFDA to PSTN phone from PSTN calls phone 1 and should CFDA to PSTN phone 1. Confirm 2-way voice 2. Confirm phone 1 number and display at PSTN phone	Pass
<b>Voicemail</b>		

TELUS_TC1.25	IP PBX phone 1 calls PSTN phone, Don't answer the call in the PSTN phone; after 4 ring, voicemail kick in Record a message Follow the prompt to play back the message Follow the prompt to cancel the recording then hang up.	Pass
<b>Conference call</b>		
TELUS_TC1.26	PSTN phone calls IP PBX phone 1 IP PBX phone 1 performs a conference call with IP PBX phone 2 Confirm audio among the parties	Pass
TELUS_TC1.27	PSTN phone calls IP PBX phone 1 IP PBX phone 1 performs a conference call with PSTN Confirm audio with PSTN phone and IP PBX phone	Pass
<b>DTMF</b>		
TELUS_TC1.28	From PBX dial 8773539586 and code 3369709( conference bridge) When hearing the prompt, enter valid Telus conference code Follow prompts and verify connected to conference bridge. Verify that pressed keys are recognized and successfully accessed conference bridge. Verify by calling to conference bridge from PSTN. Test Inband DTMF by programming PBX end point	Pass
TELUS_TC1.29	From PBX dial 8773539586 and code 3369709 ( conference bridge) When hearing the prompt, enter valid Telus conference code. Verify that pressed keys are recognized and successfully accessed conference bridge. Verify by calling to conference bridge from PSTN. Test RFC2833 by programming PBX endpoint	Pass
<b>Long calls - minimum recommendation</b>		
TELUS_TC1.31	long duration call: 10 mins - to PSTN phone	Pass
TELUS_TC1.32	long duration call on hold: Call to PSTN, PBX places call on hold for 10 min, resume call, verify 2 way audio	Pass

## VoIP Test cases

Test Number	Test Details	Pass/Fail/NA - Not Applicable
<b>Test with TELUS VoIP Account</b>		
<b>Basic inbound/outbound call</b>		
TELUS_TC2.1	Test by G.729. Call from TELUS VoIP client to IP PBX phone, 1. Confirm 2-way voice 2. Confirm the proper calling number is shown 3. Confirm the proper call display name is shown	Pass
TELUS_TC2.2	Test by setup the call with G.729. Call from IP PBX phone to TELUS VoIP client, 1. Confirm 2-way voice 2. Confirm the proper calling number is shown 3. Confirm the proper call display name is shown	Pass
<b>Basic inbound/outbound call with privacy</b>		
TELUS_TC2.3	Call from TELUS VoIP client with G.711 to IP PBX phone with privacy 1. Confirm 2-way voice 2. Confirm the proper calling number is not shown 3. Confirm the proper call display name is not shown	Pass
TELUS_TC2.4	Call from IP PBX phone to TELUS VoIP client G.711, when dialling from the IP PBX phone, use the prefix if applicable to temporary suppress the call display 1. Confirm 2-way voice 2. Confirm the proper calling number is not shown 3. Confirm the proper call display name is not shown	Pass
<b>Hold and resume</b>		
TELUS_TC2.5	Call from TELUS VoIP to IP PBX - after the call setup the PBX phone puts the call on-hold or (MOH), waits 30 seconds, resumes. Confirm audio both way after resume.	Pass
TELUS_TC2.6	Call from IP PBX to TELUS VoIP - after the call setup, use TELUS VoIP to put the call on-hold or (MOH), waits 30 seconds, resumes. Confirm 2-way voice after resume.	Pass
<b>Call Transfer (Blind transfer)</b>		
TELUS_TC2.7	IP PBX phone 1 calls IP PBX phone 2 IP PBX phone 2 performs a blind transfer to TELUS VoIP client Confirm 2-way voice after the transfer	Pass
TELUS_TC2.9	TELUS VoIP client calls IP PBX phone 1 IP PBX phone 1 performs a blind transfer to PSTN Confirm 2-way voice after the transfer	Pass
<b>Call Transfer (Consult transfer)</b>		
TELUS_TC2.11	IP PBX phone 1 calls IP PBX phone 2 IP PBX phone 2 performs a consult transfer to TELUS VoIP client Confirm 2-way voice after the transfer	Pass

TELUS_TC2.13	TELUS VoIP client calls IP PBX phone 1 IP PBX phone 1 performs a consult transfer to PSTN Confirm 2-way voice after the transfer	Pass
<b>Call Forwarding Unconditional</b>		
TELUS_TC2.15	Configure IP PBX phone 1 to CFU to TELUS VoIP client IP PBX phone 2 calls phone 1 and should CFU to TELUS VoIP client 1. Confirm 2-way voice 2. Confirm phone 1 number and display at TELUS VoIP client	Pass
TELUS_TC2.16	Configure IP PBX phone 1 to CFU to 18773539586 TELUS VoIP client calls phone 1 to trigger the call forwarding 1. Confirm 2-way voice 2. Press 1234# to interrupt the prompt	Pass
<b>Voicemail</b>		
TELUS_TC2.17	Test with G.711. IP PBX phone 1 calls TELUS VoIP client, Don't answer the call in the TELUS VoIP client; after 4 ring, voicemail kick in Record a message Follow the prompt to play back the message Follow the prompt to cancel the recording then hang up	Pass
<b>Conference call</b>		
TELUS_TC2.18	TELUS VoIP client calls IP PBX phone 1 IP PBX phone 1 performs a conference call with IP PBX phone 2 Confirm audio among the parties	Pass

## Mobile Test cases

Test Number	Test Details	Pass/Fail/NA - Not Applicable
<b>Test with TELUS mobile</b>		
<b>Basic inbound/outbound call</b>		
TELUS_TC3.1	Call from TELUS mobile client to IP PBX phone 1. Confirm 2-way voice 2. Confirm the proper calling number is shown 3. Confirm the proper call display name is shown	Pass
TELUS_TC3.2	Repeat the test by setup the call with G.729. Call from IP PBX phone to TELUS mobile client 1. Confirm 2-way voice 2. Confirm the proper calling number is shown 3. Confirm the proper call display name is shown	Pass
<b>Basic inbound/outbound call with privacy</b>		
TELUS_TC3.3	Call from TELUS mobile client to IP PBX phone G.711 with privacy enabled. 1. Confirm 2-way voice 2. Confirm the proper calling number is not shown 3. Confirm the proper call display name is not shown	Pass
TELUS_TC3.4	Call from IP PBX phone G.711 to TELUS mobile client, when dialling from the IP PBX phone, use the prefix if applicable to temporary suppress the call display 1. Confirm 2-way voice 2. Confirm the proper calling number is not shown 3. Confirm the proper call display name is not shown	Pass
<b>Hold and resume</b>		
TELUS_TC3.5	Call from TELUS mobile to IP PBX - after the call setup the PBX phone puts the call on-hold or (MOH), waits 30 seconds, resumes. Confirm audio both way after resume.	Pass
TELUS_TC3.6	Call from IP PBX to TELUS mobile - after the call setup, use TELUS mobile to put the call on-hold or (MOH), waits 30 seconds, resumes. Confirm 2-way voice after resume.	Pass
<b>Call Transfer (Blind transfer)</b>		
TELUS_TC3.7	IP PBX phone 1 calls IP PBX phone 2 IP PBX phone 2 performs a blind transfer to telus mobile client Confirm 2-way voice after the transfer	Pass
TELUS_TC3.9	TELUS mobile client calls IP PBX phone 1 IP PBX phone 1 performs a blind transfer to IP PBX phone 2 Confirm 2-way voice after the transfer	Pass

TELUS_TC3.11	TELUS mobile client calls IP PBX phone 1 IP PBX phone 1 performs a blind transfer to PSTN Confirm 2-way voice after the transfer	Pass
TELUS_TC3.13	TELUS mobile client calls IP PBX phone 1 IP PBX phone 1 performs a blind transfer to Telus mobile client --- Confirm 2-way voice after the transfer	Pass
<b>Call Transfer (Consult transfer)</b>		
TELUS_TC3.15	IP PBX phone 1 calls IP PBX phone 2 IP PBX phone 2 performs a consult transfer to Telus mobile client --- Confirm 2-way voice after the transfer	Pass
TELUS_TC3.17	TELUS mobile client calls IP PBX phone 1 IP PBX phone 1 performs a consult transfer to IP PBX phone 2 --- Confirm 2-way voice after the transfer	Pass
TELUS_TC3.19	TELUS mobile client calls IP PBX phone 1 IP PBX phone 1 performs a consult transfer to BVoIP Confirm 2-way voice after the transfer	Pass
TELUS_TC3.21	TELUS mobile client calls IP PBX phone 1 IP PBX phone 1 performs a blind transfer to another Telus Mobile client --- Confirm 2-way voice after the transfer	Pass
<b>Call Forwarding Don't Answer</b>		
TELUS_TC3.23	Configure a Mobile Phone to Forward calls to a PSTN when Dont Answer. Test G711 Mobile Phone to CFNA to TELUS PSTN Number IP PBX phone 1 calls Mobile Phone and should CFNA to TELUS PSTN Number 1. Confirm 2-way voice 2. Confirm phone 1 number and display at PSTN number	Pass
TELUS_TC3.24	Configure a Mobile Phone to Forward calls to a PSTN when Dont Answer. Test G729 Mobile Phone to CFNA to TELUS PSTN Number IP PBX phone 1 calls Mobile Phone and should CFNA to TELUS PSTN Number 1. Confirm 2-way voice 2. Confirm phone 1 number and display at PSTN number	Pass
<b>Call Forwarding Unconditional</b>		
TELUS_TC3.25	Configure IP PBX phone 1 to CFU to TELUS mobile client IP PBX phone 2 calls phone 1 and should CFU to TELUS mobile client 1. Confirm 2-way voice 2. Confirm phone 1 number and display at TELUS mobile client	Pass
<b>Voicemail</b>		



TELUS_TC3.26	Repeat the test by setup the call with G.711.Test with Movius VM platforms. IP PBX phone 1 calls TELUS mobile client Don't answer the call in the TELUS mobile client; after 4 ring, voicemail kick in Record a message Follow the prompt to play back the message Follow the prompt to cancel the recording then hang up	Pass
TELUS_TC3.27	Repeat the test by setup the call with G.729.Test with Movius VM platforms. IP PBX phone 1 calls TELUS mobile client Don't answer the call in the TELUS mobile client; after 4 ring, voicemail kick in Record a message Follow the prompt to play back the message Follow the prompt to cancel the recording then hang up	Pass
TELUS_TC3.28	From UMTS call PBX phone, CFB to VM or CFDA to PBX VM Record a message Follow the prompt to play back the message Follow the prompt to cancel the recording then hang up	Pass
<b>Conference call</b>		
TELUS_TC3.29	TELUS mobile client calls IP PBX phone 1 IP PBX phone 1 performs a conference call with IP PBX phone 2 Confirm audio among the parties	Pass
TELUS_TC3.30	TELUS mobile client calls IP PBX phone 1 IP PBX phone 1 performs a conference call with Telus VOIP Confirm audio with mobile client and IP PBX phone	Pass
TELUS_TC3.31	TELUS mobile client calls IP PBX phone 1 IP PBX phone 1 performs a conference call with PSTN Confirm audio with mobile client and IP PBX phone	Pass
TELUS_TC3.32	IP PBX phone 1 calls TELUS mobile client client a conference call to Telus VOIP Confirm audio with mobile client and IP PBX phone	Pass

## VoIP Test cases

Test Number	Test Details	Pass/Fail/NA
<b>Test with TELUS VoLTE/IMS</b>		
<b>Basic inbound/outbound call</b>		
TELUS_TC4.1	Repeat the test by setup the call with G.711. Call from IP PBX phone to TELUS VoLTE client 1. Confirm 2-way voice 2. Confirm the proper calling number is shown 3. Confirm the proper call display name is shown	Pass
TELUS_TC4.2	Repeat the test by setup the call with G.729. Call from IP PBX phone to TELUS VoLTE client 1. Confirm 2-way voice 2. Confirm the proper calling number is shown 3. Confirm the proper call display name is shown	Pass
<b>Basic inbound/outbound call with privacy</b>		
TELUS_TC4.3	Call from TELUS VoLTE client to IP PBX phone with privacy enabled. 1. Confirm 2-way voice 2. Confirm the proper calling number is not shown 3. Confirm the proper call display name is not shown	Pass
TELUS_TC4.4	Call from IP PBX phone to TELUS VoLTE client, when dialling from the IP PBX phone, use the prefix if applicable to temporary suppress the call display 1. Confirm 2-way voice 2. Confirm the proper calling number is not shown 3. Confirm the proper call display name is not shown	Pass
<b>Hold and resume</b>		
TELUS_TC4.5	Call from TELUS VoLTE to IP PBX - after the call setup the PBX phone puts the call on-hold or (MOH), waits 30 seconds, resumes. Confirm audio both way after resume.	Pass
TELUS_TC4.6	Call from IP PBX to TELUS VoLTE - after the call setup, use TELUS VoLTE to put the call on-hold or (MOH), waits 30 seconds, resumes. Confirm 2-way voice after resume.	Pass
<b>Call Transfer (Blind transfer)</b>		
TELUS_TC4.7	IP PBX phone 1 calls IP PBX phone 2 IP PBX phone 2 performs a blind transfer to telus VoLTE client Confirm 2-way voice after the transfer	Pass
TELUS_TC4.9	TELUS VoLTE client calls IP PBX phone 1 IP PBX phone 1 performs a blind transfer to IP PBX phone 2 Confirm 2-way voice after the transfer	Pass

TELUS_TC4.11	TELUS VoLTE client calls IP PBX phone 1 IP PBX phone 1 performs a blind transfer to PSTN Confirm 2-way voice after the transfer	Pass
TELUS_TC4.13	TELUS VoLTE client calls IP PBX phone 1 IP PBX phone 1 performs a blind transfer to TELUS Mobile Confirm 2-way voice after the transfer	Pass
TELUS_TC4.15	TELUS VoLTE client calls IP PBX phone 1 IP PBX phone 1 performs a blind transfer to Telus VoLTE client Confirm 2-way voice after the transfer	Pass
<b>Call Transfer (Consult transfer)</b>		
TELUS_TC4.17	IP PBX phone 1 calls IP PBX phone 2 IP PBX phone 2 performs a consult transfer to Telus VoLTE client Confirm 2-way voice after the transfer	Pass
TELUS_TC4.19	TELUS VoLTE client calls IP PBX phone 1 IP PBX phone 1 performs a consult transfer to IP PBX phone 2 Confirm 2-way voice after the transfer	Pass
TELUS_TC4.21	TELUS VoLTE client calls IP PBX phone 1 IP PBX phone 1 performs a blind transfer to BVoIP Confirm 2-way voice after the transfer	Pass
TELUS_TC4.23	TELUS VoLTE client calls IP PBX phone 1 IP PBX phone 1 performs a blind transfer to another Telus VoLTE client Confirm 2-way voice after the transfer	Pass
<b>Call Forwarding Don't Answer</b>		
TELUS_TC4.25	Configure a VoLTE Phone to Forward calls to a PSTN when Dont Answer. Test G711 IP PBX phone 1 calls VoLTE Phone and should CFNA to PSTN Number 1. Confirm 2-way voice 2. Confirm phone 1 number and display at PSTN number	Pass
TELUS_TC4.26	Configure a VoLTE Phone to Forward calls to a TELUS Mobile when Dont Answer. Test G729 IP PBX phone 1 calls VoLTE Phone and should CFNA to TELUS Mobile Number 1. Confirm 2-way voice 2. Confirm phone 1 number and display at PSTN number	Pass
<b>Call Forwarding Unconditional</b>		

TELUS_TC4.27	Configure IP PBX phone 1 to CFU to TELUS VoLTE client IP PBX phone 2 calls phone 1 and should CFU to TELUS VoLTE client 1. Confirm 2-way voice 2. Confirm phone 1 number and display at TELUS VoLTE client	Pass
<b>Voicemail</b>		
TELUS_TC4.28	Repeat the test by setup the call with G.711. Test with Movius VM platforms. IP PBX phone 1 calls TELUS VoLTE client Don't answer the call in the TELUS VoLTE client; after 4 ring, voicemail kick in Record a message Follow the prompt to play back the message Follow the prompt to cancel the recording then hang up	Pass
TELUS_TC4.29	Repeat the test by setup the call with G.729. Test with Movius VM platforms. IP PBX phone 1 calls TELUS VoLTE client Don't answer the call in the TELUS VoLTE client; after 4 ring, voicemail kick in Record a message Follow the prompt to play back the message Follow the prompt to cancel the recording then hang up	Pass
<b>Conference call</b>		
TELUS_TC4.30	TELUS VoLTE client calls IP PBX phone 1 IP PBX phone 1 performs a conference call with IP PBX phone 2 Confirm audio among the parties	Pass
TELUS_TC4.31	TELUS VoLTE client calls IP PBX phone 1 IP PBX phone 1 performs a conference call with Telus VOIP Confirm audio with VoLTE client and IP PBX phone	Pass
TELUS_TC4.32	IP PBX phone 1 calls TELUS VoLTE client client a conference call to Telus VOIP Confirm audio with VoLTE client and IP PBX phone	Pass
TELUS_TC4.33	IP PBX phone 1 calls TELUS VoLTE client client a conference call to Telus Mobile Confirm audio with VoLTE client and IP PBX phone	Pass

## NGHP Test cases

Test Number	Test Details	Pass/Fail/NA - Not Applicable
<b>Test with TELUS NGHP</b>		
<b>Basic inbound/outbound call</b>		
TELUS_TC5.1	Repeat the test by setup the call with G.729. Call from IP PBX phone to TELUS NGHP client 1. Confirm 2-way voice 2. Confirm the proper calling number is shown 3. Confirm the proper call display name is shown	Pass
<b>Basic inbound/outbound call with privacy</b>		
TELUS_TC5.2	Call from TELUS NGHP client to IP PBX phone with privacy enabled. 1. Confirm 2-way voice 2. Confirm the proper calling number is not shown 3. Confirm the proper call display name is not shown	Pass
TELUS_TC5.3	Call from IP PBX phone to TELUS NGHP client, when dialling from the IP PBX phone, use the prefix if applicable to temporary suppress the call display 1. Confirm 2-way voice 2. Confirm the proper calling number is not shown 3. Confirm the proper call display name is not shown	Pass
<b>Hold and resume</b>		
TELUS_TC5.4	Call from TELUS NGHP to IP PBX - after the call setup the PBX phone puts the call on-hold or (MOH), waits 30 seconds, resumes. Confirm audio both way after resume.	Pass
TELUS_TC5.5	Call from IP PBX to TELUS NGHP - after the call setup, use TELUS NGHP to put the call on-hold or (MOH), waits 30 seconds, resumes. Confirm 2-way voice after resume.	Pass
<b>Call Transfer (Blind transfer)</b>		
TELUS_TC5.6	IP PBX phone 1 calls IP PBX phone 2 IP PBX phone 2 performs a blind transfer to telus NGHP client Confirm 2-way voice after the transfer	Pass
TELUS_TC5.8	TELUS NGHP client calls IP PBX phone 1 IP PBX phone 1 performs a blind transfer to IP PBX phone 2 Confirm 2-way voice after the transfer	Pass
TELUS_TC5.10	TELUS NGHP client calls IP PBX phone 1 IP PBX phone 1 performs a blind transfer to BVoIP Confirm 2-way voice after the transfer	Pass
TELUS_TC5.12	TELUS NGHP client calls IP PBX phone 1 IP PBX phone 1 performs a blind transfer to TELUS Mobile Confirm 2-way voice after the transfer	Pass

TELUS_TC5.14	TELUS NGHP client calls IP PBX phone 1 IP PBX phone 1 performs a blind transfer to VoLTE Confirm 2-way voice after the transfer	Pass
TELUS_TC5.16	TELUS NGHP client calls IP PBX phone 1 IP PBX phone 1 performs a blind transfer to Telus NGHP client Confirm 2-way voice after the transfer	Pass
<b>Call Transfer (Consult transfer)</b>		
TELUS_TC5.18	IP PBX phone 1 calls IP PBX phone 2 IP PBX phone 2 performs a consult transfer to Telus NGHP client Confirm 2-way voice after the transfer	Pass
TELUS_TC5.20	TELUS NGHP client calls IP PBX phone 1 IP PBX phone 1 performs a consult transfer to IP PBX phone 2 Confirm 2-way voice after the transfer	Pass
TELUS_TC5.22	TELUS NGHP client calls IP PBX phone 1 IP PBX phone 1 performs a consult transfer to BVoIP Confirm 2-way voice after the transfer	Pass
TELUS_TC5.24	TELUS NGHP client calls IP PBX phone 1 IP PBX phone 1 performs a consult transfer to TELUS Mobile Confirm 2-way voice after the transfer	Pass
TELUS_TC5.26	TELUS NGHP client calls IP PBX phone 1 IP PBX phone 1 performs a consult transfer to VoLTE Confirm 2-way voice after the transfer	Pass
TELUS_TC5.28	TELUS NGHP client calls IP PBX phone 1 IP PBX phone 1 performs a blind transfer to another Telus NGHP client Confirm 2-way voice after the transfer	Pass
<b>Call Forwarding Unconditional</b>		
TELUS_TC5.33	Configure a NGHP Phone to Forward calls to a PSTN unconditional IP PBX phone 1 calls NGHP Phone and should CFU to PSTN Number 1. Confirm 2-way voice 2. Confirm phone 1 number and display at PSTN number	Pass
TELUS_TC5.34	Configure a NGHP Phone to Forward calls to a VoLTE unconditional IP PBX phone 1 calls NGHP Phone and should CFU toVoLTE Number 1. Confirm 2-way voice 2. Confirm phone 1 number and display at PSTN number	Pass

TELUS_TC5.35	Configure a NGHP Phone to Forward calls to a TELUS Mobile unconditional. IP PBX phone 1 calls NGHP Phone and should CFU to TELUS Mobile Number 1. Confirm 2-way voice 2. Confirm phone 1 number and display at PSTN number	Pass
<b>Voicemail</b>		
TELUS_TC5.36	Repeat the test by setup the call with G.711.Test with Movius VM platforms. IP PBX phone 1 calls TELUS NGHP client Don't answer the call in the TELUS NGHP client; after 4 ring, voicemail kick in Record a message Follow the prompt to play back the message Follow the prompt to cancel the recording then hang up	Pass
TELUS_TC5.37	Repeat the test by setup the call with G.729.Test with Movius VM platforms. IP PBX phone 1 calls TELUS NGHP client Don't answer the call in the TELUS NGHP client; after 4 ring, voicemail kick in Record a message Follow the prompt to play back the message Follow the prompt to cancel the recording then hang up	Pass
<b>Conference call</b>		
TELUS_TC5.38	TELUS NGHP client calls IP PBX phone 1 IP PBX phone 1 performs a conference call with IP PBX phone 2 Confirm audio among the parties	Pass
TELUS_TC5.39	TELUS NGHP client calls IP PBX phone 1 IP PBX phone 1 performs a conference call with Telus VOIP Confirm audio with NGHP client and IP PBX phone	Pass
TELUS_TC5.40	TELUS NGHP client calls IP PBX phone 1 IP PBX phone 1 performs a conference call with VoLTE Confirm audio with NGHP client and IP PBX phone	Pass
TELUS_TC5.41	IP PBX phone 1 calls TELUS NGHP client client a conference call to Telus Mobile Confirm audio with NGHP client and IP PBX phone	Pass

## Troubleshooting Tools

### Wireshark

Wireshark is also a network protocol analyzer which is freely downloadable from [www.wireshark.org](http://www.wireshark.org).  
On the Oracle E-SBC

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

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

At the E-SBC Console:

```
SBC1# reset sipd
SBC1# notify sipd debug
SBC1#
enabled SIP Debugging
SBC1# notify all rotate-logs
```

#### **Examining the log files**

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

```
C:\Documents and Settings\user>ftp 192.168.1.22
Connected to 192.168.85.55.
220 SBC1 server (VxWorks 6.4) ready. User (192.168.1.22:(none)): user
331 Password required for user. Password: acme
230 User user logged in.
ftp> cd /opt/logs
250 CWD command successful. ftp> get sipmsg.log
200 PORT command successful.
150 Opening ASCII mode data connection for '/opt/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 '/opt/logs/log.sipd' (204681
bytes).
226 Transfer complete.
ftp: 206823 bytes received in 0.11Seconds 1897.46Kbytes/sec
```

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

#### **Through the Web GUI**

You can also check the display results of filtered SIP session data from the Oracle Enterprise Session Border Controller, and provides traces in a common log format for local viewing or for exporting to your PC. Please check the “Monitor and Trace” section (page 145) of the Web GUI User Guide available at [http://docs.oracle.com/cd/E56581\\_01/index.htm](http://docs.oracle.com/cd/E56581_01/index.htm)



## Appendix A

### Full E-SBC Configuration

```
filter-config
  name          all
  user          *
local-policy
  from-address  *
  to-address    *
  source-realm  Core
  next-hop      172.25.128.75
  realm         Peer
  action        replace-uri
local-policy
  from-address  *
  to-address    *
  source-realm  Peer
  next-hop      172.24.0.162
  realm         Core
  action        replace-uri
media-manager
  media-policing disabled
network-interface
  name          M00
  ip-address    198.162.151.37
  netmask       255.255.255.0
  gateway       198.162.151.38
  hip-ip-list   198.162.151.37
  icmp-address  198.162.151.37
network-interface
  name          M10
  ip-address    172.24.0.165
  netmask       255.255.255.0
  gateway       172.24.0.254
  hip-ip-list   172.24.0.165
  icmp-address  172.24.0.165
  ssh-address   172.24.0.165
phy-interface
  name          M00
  operation-type Media
phy-interface
  name          M10
  operation-type Media
  slot          1
phy-interface
  name          wancom0
  duplex-mode
  speed
realm-config
  identifier     Core
  network-interfaces M10:0
realm-config
  identifier     Peer
  network-interfaces M00:0
session-agent
  hostname       172.24.0.162
  ip-address     172.24.0.162
  transport-method StaticTCP
  realm-id       Core
  description    CUCM 8.6
  ping-method    OPTIONS;hops=0
  ping-interval  60
session-agent
  hostname       172.25.128.75
  ip-address     172.25.128.75
  realm-id       Peer
```

description	Telus SIP trunk
in-manipulationid	towardstrunk
sip-config	
registrar-domain	*
registrar-host	*
registrar-port	5060
options	max-udp-length=0
extra-method-stats	enabled
sip-feature	
name	100rel
realm	Peer
require-mode-inbound	Pass
require-mode-outbound	Pass
sip-interface	
realm-id	Core
sip-port	
address	172.24.0.165
transport-protocol	TCP
sip-port	
address	172.24.0.165
options	100rel-interworking
sip-interface	
realm-id	Peer
sip-port	
address	198.162.151.37
options	100rel-interworking
sip-manipulation	
name	towardstrunk
header-rule	
name	DelReq
header-name	Require
action	delete
match-value	100rel
sip-monitoring	
monitoring-filters	*
steering-pool	
ip-address	172.24.0.165
start-port	50000
end-port	60000
realm-id	Core
steering-pool	
ip-address	198.162.151.37
start-port	50000
end-port	60000
realm-id	Peer
system-config	
enable-snmp-monitor-traps	enabled
enable-env-monitor-traps	enabled
syslog-server	
address	172.17.71.8
syslog-server	
address	172.18.129.141
syslog-server	
address	172.18.129.144
syslog-server	
address	172.18.129.154
system-log-level	MAJOR
process-log-level	MAJOR
comm-monitor	
state	enabled
monitor-collector	
address	172.18.159.108
call-trace	enabled
default-gateway	172.17.100.254
source-routing	enabled
trap-receiver	
ip-address	172.17.71.8:162
filter-level	All
trap-receiver	

ip-address	172.18.129.141:162
filter-level	Major
trap-receiver	
ip-address	172.18.129.144:162
filter-level	Major
trap-receiver	
ip-address	172.18.129.154:162
filter-level	Major
web-server-config	

## 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.

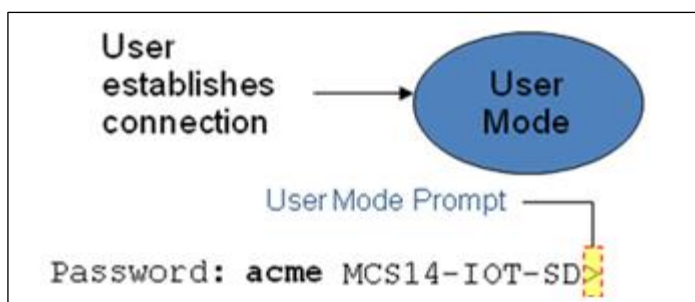
Initial connectivity will be through the serial console port. At a minimum, this is how to configure the management (eth0) interface on the SBC.

### ACLI Basics

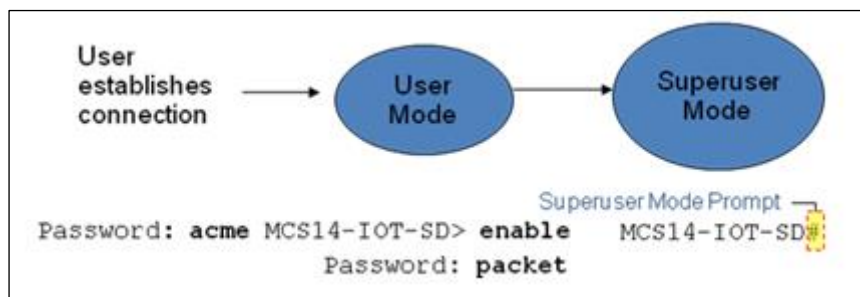
There are two password protected modes of operation within the ACLI, User mode and Superuser mode.

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.



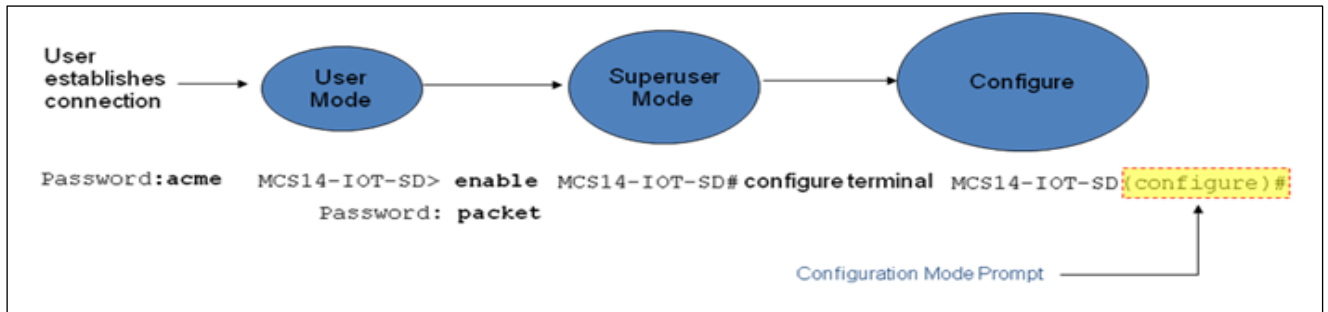
The Superuser mode allows for access to all system commands for operation, maintenance, and administration. This mode is identified by the pound sign (#) in the prompt after the target name. To enter the Superuser mode, issue the enable command in the User mode.



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.

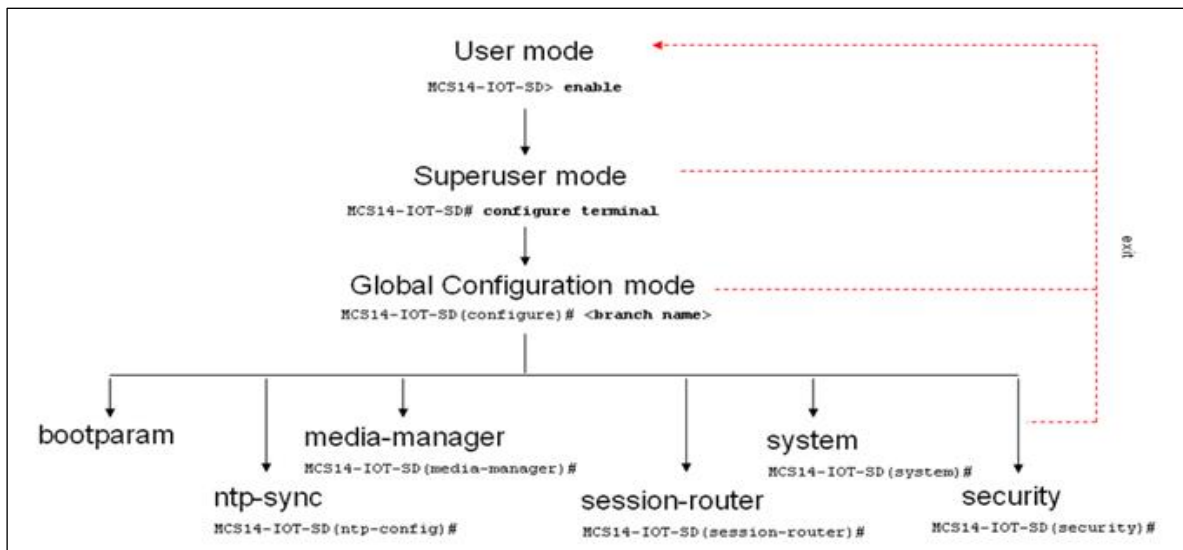
You must enter the Configuration mode to configure elements. For example, you can access the configuration branches and configuration elements for signaling and media configurations. To enter the Configuration mode, issue the configure terminal command in the Superuser mode.

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, iwf-config, local-policy, sip-manipulation, session-agent, etc.

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

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

## **Configuration Elements**

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

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

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

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

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

## **Creating an Element**

1. To create a single-instance element, you go to the appropriate level in the ACLI path and enter its parameters. There is no need to specify a unique identifier property because a single-instance element is a global element and there is only one instance of this element.
2. When creating a multiple-instance element, you must specify a unique identifier for each instance of the element.
3. It is important to check the parameters of the element you are configuring before committing the changes. You do this by issuing the show command before issuing the done command. The parameters that you did not configure are filled with either default values or left empty.
4. On completion, you must issue the done command. The done command causes the configuration to be echoed to the screen and commits the changes to the volatile memory. It is a good idea to review this output to ensure that your configurations are correct.
5. Issue the exit command to exit the selected element.

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

## **Editing an Element**

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

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

5. On completion, you must issue the done command.
6. Issue the exit command to exit the selected element.

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

### **Deleting an Element**

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

To delete a single-instance element,

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

To delete a multiple-instance element,

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

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

### **Configuration Versions**

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

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

To view the running configuration, issue command show running-config.

### **Saving the Configuration**

The save-config command stores the edited configuration persistently.

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

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

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

```
oraclesbc1 # 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'.
oraclesbc1 #
```

## 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.



```
oraclesbc1# activate-config
Activate-Config received, processing.
waiting 120000 for request to finish
Request to 'ACTIVATE-CONFIG' has Finished,
Activate Complete
oraclesbc1#
```



Oracle Corporation, World Headquarters Worldwide Inquiries

500 Oracle Parkway Phone: +1.650.506.7000  
Redwood Shores, CA 94065, USA Fax: +1.650.506.7200

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