

**Hardware and Software**  
**Engineered to Work Together**



Oracle Enterprise Session Border Controller  
and Microsoft Lync 2013 with Telus Enterprise  
SIP Trunking R2 - Dedicated Line.

Technical Application Note

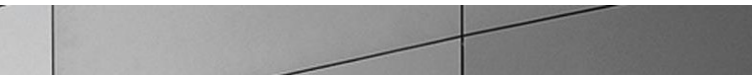


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## Intended Audience

This document is intended for use by Oracle Systems Engineers, 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.

## Document Overview

Microsoft Lync 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 Microsoft Lync, Oracle E-SBCs and IP Trunking services are configured in the optimal manner.

## Introduction

## Audience

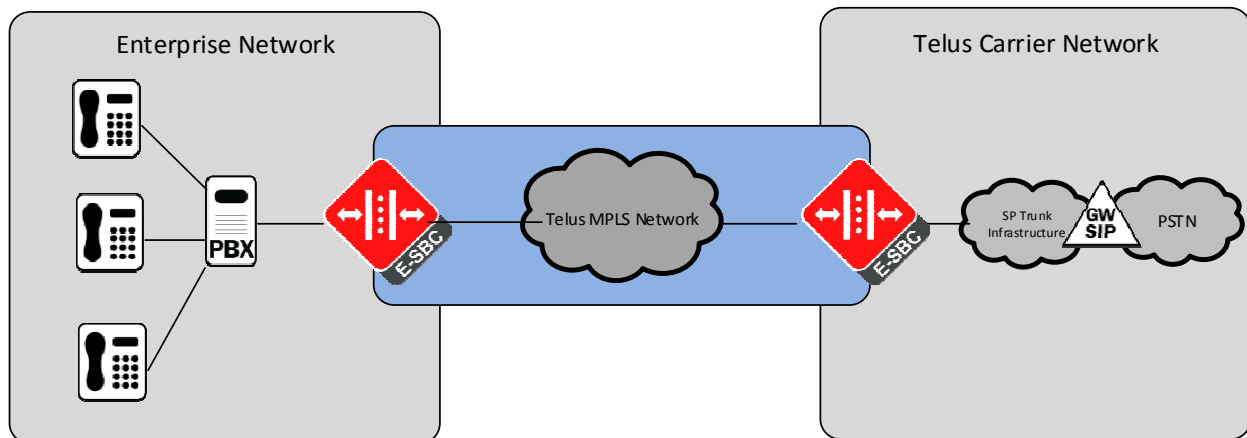
This is a technical document intended for telecommunications engineers with the purpose of configuring the Oracle Enterprise Session Border Controller and Microsoft Lync. There will be steps that require navigating the Command Line Interface (CLI). Understanding the basic concepts of TCP/UDP, IP/Routing,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

## Requirements

- Microsoft Lync 2013 – cumulative update 5.0.8308.577
- Oracle Enterprise Session Border Controller is running ECZ720p2.64.bz. (Build 166) Note: the configuration running on the E-SBC is backward/forward compatible with any release in the 7.2.0 stream.
- Telus trunk based customers with dedicated data connectivity to Telus.

## Architecture

The following reference architecture shows a logical view of the connectivity between CM and the E-SBC.



## Lab Configuration

Following are the IP addresses used for the Interoperability tests. The IPs below are specific to lab setup at Telus, the IPs in production will be vastly different from network addresses listed below.

description	network-interface	realm	interface IP	Host Name	sip-port
SBC interfaces					
management	wancom0		192.168.1.22		
redundancy	wancom1		169.254.1.1		
redundancy	wancom2		169.254.2.1		
media/signaling	s0p0:0	inside	172.16.153.34	lync-acme-sbc.teluscpslynclab.net	5067
media/signaling	s1p0:0	outside	172.16.154.35		5067
Session-Agents					
Lync Mediation Server 1		inside	172.16.149.38	fe0101.teluscpslynclab.net	5066
Lync Mediation Server 2		inside	172.16.149.39	fe0102.teluscpslynclab.net	5066
Lync Mediation Server 3		inside	172.16.149.40	fe0103.teluscpslynclab.net	5066
Telus trunk		outside	10.27.56.7		5060

## 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 CM 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 see 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 ECZ7.2.0 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 CM 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

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        : /code/images/nnECZ720p2.64.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.

### High Availability

For additional information on High Availability please see the enterprise SBC documentation for more information (<http://www.oracle.com/technetwork/indexes/documentation/oracle-comms-acme-packet-2046907.html>)

Interfaces wancom1 and 2 need to be added to facilitate HA communication between the two HA pairs.

```

network-interface
  name                wancom1
  sub-port-id         0
  description         HA_HEARTBEAT1
  hostname
  ip-address
  pri-utility-addr    169.254.1.1
  sec-utility-addr    169.254.1.2
  netmask             255.255.255.252
  gateway
  sec-gateway
  gw-heartbeat
    state             disabled
    heartbeat         0
    retry-count       0
    retry-timeout     1
    health-score      0
  dns-ip-primary
  dns-ip-backup1
  dns-ip-backup2
  dns-domain
  dns-timeout         11
  hip-ip-list
  ftp-address
  icmp-address
  snmp-address
  telnet-address
  ssh-address
network-interface
  name                wancom2
  sub-port-id         0
  description         HA_HEARTBEAT2
  hostname

```

```

ip-address
pri-utility-addr          169.254.2.1
sec-utility-addr         169.254.2.2
netmask                   255.255.255.252
gateway
sec-gateway
gw-heartbeat
    state                  disabled
    heartbeat              0
    retry-count            0
    retry-timeout          1
    health-score           0
dns-ip-primary
dns-ip-backup1
dns-ip-backup2
dns-domain
dns-timeout               11
hip-ip-list
ftp-address
icmp-address
snmp-address
telnet-address
ssh-address

```

```

redundancy-config
    becoming-standby-time  360000
peer
    name                   SBC1
    type                   Primary
    destination
        address            169.254.1.1:9090
        network-interface  wancom1:0
    destination
        address            169.254.2.1:9090
        network-interface  wancom2:0
peer
    name                   SBC2
    type                   Secondary
    destination
        address            169.254.1.2:9090
        network-interface  wancom1:0
    destination
        address            169.254.2.2:9090
        network-interface  wancom2:0

```

Additionally primary and secondary interface IPs need to be added to the media/signaling network-interfaces

```

network-interface
    name                   s0p0
    sub-port-id            0
    description            Outside/Untrusted
    hostname
    ip-address             172.16.153.34
    pri-utility-addr       172.16.153.2
    sec-utility-addr       172.16.153.3
    netmask                 255.255.255.0
    gateway                 172.16.153.1

```

```

sec-gateway
gw-heartbeat
    state                disabled
    heartbeat            0
    retry-count          0
    retry-timeout        1
    health-score         0
dns-ip-primary
dns-ip-backup1
dns-ip-backup2
dns-domain
dns-timeout              11
signaling-mtu            0
hip-ip-list
ftp-address
icmp-address
snmp-address
telnet-address
ssh-address
network-interface
name                     slp0
sub-port-id              0
description              Inside/Trusted
hostname
ip-address               172.16.154.35
pri-utility-addr         172.16.154.2
sec-utility-addr         172.16.154.3
netmask                  255.255.255.0
gateway                  172.16.154.1
sec-gateway
gw-heartbeat
    state                disabled
    heartbeat            0
    retry-count          0
    retry-timeout        1
    health-score         0
dns-ip-primary
dns-ip-backup1
dns-ip-backup2
dns-domain
dns-timeout              11
signaling-mtu            0
hip-ip-list
ftp-address
icmp-address
snmp-address
telnet-address
ssh-address

```

## Routing via Local Policy

For outbound calls the local-policy determines which trunk to forward the call based on the NPA of the request-URI. This is configured in the local policy of the "To". For most configurations there will be only 1 inside and outside realm. For a single inside/outside realm configuration the local policy to and from would be set to "\*". Redundant trunk configurations will use a session-agent group.

```

local-policy
  from-address *
  to-address *
  source-realm outside
  description
  activate-time
  deactivate-time
  state enabled
  policy-priority none
  policy-attribute
    next-hop SAG:med-grp-1
    realm inside
    action none
    terminate-recursion disabled
    carrier
    start-time 0000
    end-time 2400
    days-of-week U-S
    cost 0
    state enabled
    app-protocol SIP
    methods
    media-profiles
    lookup single
    next-key
    eloc-str-lkup disabled
    eloc-str-match

```

```

local-policy
  from-address *
  to-address *
  source-realm inside
  description
  activate-time
  deactivate-time
  state enabled
  policy-priority none
  policy-attribute
    next-hop SAG:med-grp-1
    realm peer
    action none
    terminate-recursion disabled
    carrier
    start-time 0000
    end-time 2400
    days-of-week U-S
    cost 0
    state enabled
    app-protocol SIP
    methods
    media-profiles
    lookup single
    next-key
    eloc-str-lkup disabled
    eloc-str-match

```

```

session-group
  group-name med-grp-1
  description Lync Mediation server group

```

```

state enabled
app-protocol SIP
strategy Hunt
dest fe0101.teluscpslynclab.net
      fe0102.teluscpslynclab.net
      fe0103.teluscpslynclab.net

trunk-group
sag-recursion disabled
stop-sag-recurse 401,407

```

## Header manipulation rules required for the Telus Trunk

The HMRs update the host portion of the URI to the Telus trunk IP for Request-URI and To headers. The host portion of the URI is updated with the E-SBC outside sip-interface IP for From, P-Asserted-Identity and Contact so that the E-SBC presents its interface IP to the next hop.

```

header-rule
    name save PAI
    header-name P-Asserted-Identity
    action store
    comparison-type case-sensitive
    msg-type any
    methods
    match-value
    new-value
header-rule
    name Updt PAI
    header-name P-Asserted-Identity
    action add
    comparison-type boolean
    msg-type any
    methods INVITE
    match-value !$save PAI
    new-value <sip:
2223334444@ipinet4.com;user=phone>
header-rule
    name Updt RURI
    header-name request-uri
    action manipulate
    comparison-type case-sensitive
    msg-type any
    methods
    match-value
    new-value
element-rule
    name
Udpt URI Host
    parameter-name
    type uri-host
    action replace
    match-val-type any
    comparison-type case-
sensitive
    match-value
    new-value ipinet4.com

```

	header-rule		
	name		Updt To
	header-name		To
	action		manipulate
	comparison-type		case-sensitive
	msg-type		any
	methods		
	match-value		
	new-value		
	element-rule		
	name		
UPdt URI host		parameter-name	
		type	uri-host
		action	replace
		match-val-type	any
sensitive		comparison-type	case-
		match-value	
		new-value	ipinet4.com
	header-rule		
	name		Updt From
	header-name		From
	action		manipulate
	comparison-type		case-sensitive
	msg-type		any
	methods		
	match-value		
	new-value		
	element-rule		
	name		
Updt URI host		parameter-name	
		type	uri-host
		action	replace
		match-val-type	any
sensitive		comparison-type	case-
		match-value	
		new-value	ipinet4.com
	header-rule		
	name		Updt Contact
	header-name		Contact
	action		manipulate
	comparison-type		case-sensitive
	msg-type		any
	methods		
	match-value		
	new-value		
	element-rule		
	name		
Updt URI Host		parameter-name	
		type	uri-host
		action	replace
		match-val-type	any
		comparison-type	case-

```

sensitive
    match-value
    new-value                $LOCAL IP
    element-rule
        name                 Del MSOpaque
        parameter-name       ms-opaque
        type                  uri-param
        action                delete-
element
    match-val-type          any
    comparison-type         case-
sensitive
    match-value
    new-value

```

## Header manipulation rules to support privacy calling

Lync does not support privacy calling. The E-SBC can help support privacy calling through header manipulation rules. The Lync Administrator needs to support the appropriate \*-code in the dial-plan. In the provided example \*67 provides privacy. If the SBC detects \*67 as a prefix in the request URI, the SBC will apply RFC3323 (A Privacy Mechanism for the Session Initiation Protocol).

```

    header-rule
        name                 CheckPrivacy
        header-name          request-uri
        action                store
        comparison-type       case-sensitive
        msg-type              any
        methods                INVITE
        match-value
        new-value
        element-rule
            name                 CheckStar67
            parameter-name
            type                 uri-user
            action                store
            match-val-type      any
            comparison-type     pattern-
rule
    match-value              \*67\d+
    new-value
        header-rule
            name                 AddPrivacyHdr
            header-name          Privacy
            action                add
            comparison-type       boolean
            msg-type              request
            methods                INVITE
            match-value          $CheckPrivacy.$CheckStar67
            new-value            id
        header-rule
            name                 updateRURI
            header-name          request-uri
            action                manipulate
            comparison-type       pattern-rule
            msg-type              request
            methods                INVITE

```



	match-value	
	new-value	
	element-rule	
	name	
updateRURIUser		
	parameter-name	
	type	uri-user
	action	replace
	match-val-type	any
rule	comparison-type	pattern-
	match-value	\*67(.*)
	new-value	\$1
	header-rule	
	name	updateTO
	header-name	To
	action	manipulate
	comparison-type	pattern-rule
	msg-type	request
	methods	INVITE
	match-value	
	new-value	
	element-rule	
	name	updateTOusr
	parameter-name	
	type	uri-user
	action	replace
	match-val-type	any
rule	comparison-type	pattern-
	match-value	\*67(.*)
	new-value	\$1
	header-rule	
	name	StoreFromTag
	header-name	From
	action	store
	comparison-type	case-sensitive
	msg-type	request
	methods	INVITE
	match-value	
	new-value	
	element-rule	
	name	storeTag
	parameter-name	tag
param	type	header-
	action	store
	match-val-type	any
sensitive	comparison-type	case-
	match-value	
	new-value	
	header-rule	
	name	ChgFromPrivacy
	header-name	From
	action	manipulate
	comparison-type	boolean

```

msg-type request
methods INVITE
match-value
$CheckPrivacy.$CheckStar67
new-value "\"Anonymous\""
<sip:anonymous@anonymous.invalid>; taq="+$StoreFromTaq.$storeTaq.$0

```

## SRTP Configuration

SRTP provides encrypted audio streams to/from Lync to the Oracle Enterprise Session Boarder Controller. Telus Trunking does not support SRTP. For more information regarding SRTP configuration procedures please review the Enterprise Session Border Controller Configuration Guide.

```

sdes-profile
  name sdes1
  crypto-list AES_CM_128_HMAC_SHA1_80
              AES_CM_128_HMAC_SHA1_32
  srtp-auth enabled
  srtp-encrypt enabled
  srtcp-encrypt enabled
  mki disabled
  egress-offer-format simultaneous-best-effort
  use-ingress-session-params
  options
  key
  salt

```

```

media-sec-policy
  name rtponly
  pass-through disabled
  options
  inbound
    profile
    mode rtp
    protocol none
  outbound
    profile
    mode rtp
    protocol none

```

```

media-sec-policy
  name sdespolicy
  pass-through disabled
  options
  inbound
    profile sdes1
    mode srtp
    protocol sdes
  outbound
    profile sdes1
    mode srtp
    protocol sdes

```

```

realm-config
  identifier                inside
  description
  addr-prefix                0.0.0.0
  network-interfaces        s0p0:0
  mm-in-realm                disabled
  mm-in-network              enabled
  mm-same-ip                 enabled
  mm-in-system               enabled
...
  media-policy
  media-sec-policy          sdespolicy
  srtp-msm-passthrough      disabled

```

```

realm-config
  identifier                outside
  description
  addr-prefix                0.0.0.0
  network-interfaces        s0p1:0
  mm-in-realm                disabled
  mm-in-network              enabled
  mm-same-ip                 enabled
  mm-in-system               enabled
...
  media-policy
  media-sec-policy          rtponly
  srtp-msm-passthrough      disabled

```

## TLS Configuration

TLS provides encrypted SIP signaling between the Oracle Communications E-SBC and Lync 2013. TLS requires the exchange of certificates. The Lync administrator will need to provide the local domain controller root certificate. Likewise the CSR created on the E-SBC will need to be signed by the domain controller certificate authority that the mediation servers are associated with. The signed certificate will then need to be imported back into the SBC. For more information regarding TLS configuration procedures please review the Enterprise Session Border Controller Configuration Guide.

```

certificate-record
  name                      ESBCCert1
  country                     US
  state                        MA
  locality                     Burlington
  organization                  Engineering
  unit
  common-name              lync-acme-
sbc.teluscpslynclab.net
  key-size                  2048
  alternate-name
  trusted                       enabled
  key-usage-list                digitalSignature
                                   keyEncipherment
  extended-key-usage-list      serverAuth
  options

```

```

certificate-record

```

<b>name</b>	<b>MediationRoot</b>
country	US
state	MA
locality	Burlington
organization	Engineering
unit	
<b>common-name</b>	<b>teluscpslynclab-DC1-LYNCLAB-CA-1</b>
<b>key-size</b>	<b>2048</b>
alternate-name	
trusted	enabled
key-usage-list	digitalSignature keyEncipherment
extended-key-usage-list	serverAuth
options	

tls-profile	
<b>name</b>	<b>Core</b>
<b>end-entity-certificate</b>	<b>ESBCCert1</b>
<b>trusted-ca-certificates</b>	<b>MediationRoot</b>
cipher-list	ALL
verify-depth	10
<b>mutual-authenticate</b>	<b>enabled</b>
tls-version	compatibility
options	
cert-status-check	disabled
cert-status-profile-list	
ignore-dead-responder	disabled
allow-self-signed-cert	disabled

sip-interface	
state	enabled
realm-id	inside
description	
sip-port	
address	172.16.153.34
<b>port</b>	<b>5066</b>
<b>transport-protocol</b>	<b>TLS</b>
<b>tls-profile</b>	<b>Core</b>
allow-anonymous	agents-only
multi-home-addr	
ims-aka-profile	

session-agent	
hostname	fe0101.teluscpslynclab.net
ip-address	172.16.149.38
<b>port</b>	<b>5067</b>
state	enabled
app-protocol	SIP
app-type	
<b>transport-method</b>	<b>StaticTLS</b>
...	

## 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. 1) The webserver must be enabled. 2) 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

Caveats and out of scope items: Fax was not tested because the Lab CM did not have an analog card to test these capability there for Fax is considered out of scope for this testing.

Following is the test plan executed against this setup and results have been documented below.

Test Number	Test Details	Pass/Fail/NA - Not Applicable	Test results (Comments)
Basic originated and terminated calls			
TELUS_TC1	Call to following number from PBX: 647-837-0597 Toronto Local 780-429-7423 Edmonton Local 613-683-0932 Ottawa Local 514-788-7663 Montreal Local 604-681-0262 Vancouver Local 403-532-8075 Calgary Local 1-877-353-9586 Toll Free North America When hearing the prompt, press 1234# to interrupt the prompt. Should hear "invalid access code" prompt to confirm the DTMF tone detection. Note that you may need to dial 1 as it could be a LD call.	Pass	
TELUS_TC2	Call to the following test line - 9056352304. After the call is answered, you will hear a "confirmation tone" and you could disconnect the call. Note that you may need to dial 1 as it could be a LD call.	Pass	
Test with PSTN line			
Pre-test Procedure / Requirements			

Customer supplied PSTN line for test, please contact Telus for test case involving Telus mobile test line (Note 2)

Basic inbound/outbound call

TELUS_TC3	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_TC4	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	

Basic inbound/outbound call with privacy

TELUS_TC5	Call from PSTN phone to IP PBX phone, prefix the IP PBX phone number with *63 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_TC6	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	

Hold and resume

TELUS_TC7	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_TC8	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	

Call Transfer (Blind transfer)

TELUS_TC9	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 Repeat the same test using SIP REFER	Pass	
TELUS_TC10	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 Repeat the same test using SIP REFER	Pass	
TELUS_TC11	PSTN phone calls IP PBX phone 1 IP PBX phone 1 performs a blind transfer to another PSTN Confirm both way audio. Repeat the same test using SIP REFER	Pass	

TELUS_TC12	PSTN phone calls IP PBX phone 1 IP PBX phone 1 performs a blind transfer to Telus mobile client Confirm both way audios Repeat the same test using SIP REFER	Pass	
Call Transfer (Consult transfer)			
TELUS_TC13	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 Repeat the same test using SIP REFER	Pass	
TELUS_TC14	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 Repeat the same test using SIP REFER	Pass	
TELUS_TC15	PSTN phone calls IP PBX phone 1 IP PBX phone 1 performs a consult transfer to another PSTN Confirm both way audio. Repeat the same test using SIP REFER	Pass	
TELUS_TC16	PSTN phone calls IP PBX phone 1 IP PBX phone 1 performs a consult transfer to Telus mobile client Confirm both way audios Repeat the same test using SIP REFER	Pass	
Call Forwarding Unconditional			
TELUS_TC17	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_TC18	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	
TELUS_TC19	Configure IP PBX phone 1 to CFU to telus mobile client PSTN phone calls phone 1 to trigger the call forwarding 1. Confirm audio prompt 2. confirm the phone 1 number and display Mobile client	Pass	
Call Forwarding Busy			
TELUS_TC20	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_TC21	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	

TELUS_TC19	Configure IP PBX phone 1 to CFU to Telus Mobile client PSTN phone calls phone 1 to trigger the call forwarding 1. Confirm audio prompt 2. Press 1234# to interrupt the prompt	Pass	
Call Forwarding Don't Answer			
TELUS_TC22	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_TC23	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	
TELUS_TC19	Configure IP PBX phone 1 to CFU to Telus Mobile client PSTN phone calls phone 1 to trigger the call forwarding 1. Confirm audio prompt 2. confirm phone 1 number and display on mobile client	Pass	
Voicemail			
TELUS_TC24	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	
Conference call			
TELUS_TC25	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_TC26	PSTN phone calls IP PBX phone 1 IP PBX phone 1 performs a conference call with 1-877-353-9586 Confirm audio with PSTN phone and IP PBX phone	Pass	
Long calls - minimum recommendation			
TELUS_TC28	long duration call: 2 hours - to PSTN phone	Pass	
TELUS_TC29	long duration call on hold: Call to PSTN, PBX places call on hold for 20 min, resume call, verify 2 way audio	Pass	
TELUS_TC27	IP PBX phone 1 calls PSTN phone IP PBX phone 1 performs a conference call to Telus Mobile client Confirm audio with PSTN phone and IP PBX phone	Pass	
FAX			



TELUS_TC30	Repeat the test by setup the call with both G.711 and G.729. Outbound (from IP PBX to PSTN) T.38 testing , set up the call with G711, PBX re-invite with T38. verified the fax passed with T.38.	Not Supported	No native support for fax with Lync
TELUS_TC31	Inbound (from PSTN to IP PBX) T.38 testing	Not Supported	No native support for fax with Lync
TELUS_TC32	Repeat the test by setup the call with both G.711 and G.729. Outbound (from IP PBX to PSTN) FAX G.711 pass-through testing,test G711 fax pass through.	Not Supported	No native support for fax with Lync
TELUS_TC33	Inbound (from PSTN to IP PBX) FAX G.711 pass-through testing	Not Supported	No native support for fax with Lync
Test with TELUS VoIP Account			
Pre-test Procedure / Requirements			
Schedule test time with TELUS account representative. TELUS contact is required to support VoIP testing. TELUS IOT contact will share VoIP number and Telus Mobile client at time of test.			
Basic inbound/outbound call			
TELUS_TC27	Repeat the test by both G.711 and 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_TC28	Repeat the test by setup the call with both G.711 and 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	
Basic inbound/outbound call with privacy			
TELUS_TC29	Call from TELUS VoIP client to IP PBX phone with privacy 1. CConfirm 2-way voice 2. Confirm the proper calling number is not shown 3. Confirm the proper call display name is not shown	Pass	
TELUS_TC30	Call from IP PBX phone to TELUS VoIP 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	
Hold and resume			
TELUS_TC31	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_TC32	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	
Call Transfer (Blind transfer)			

TELUS_TC33	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 Repeat the same test using SIP REFER	Pass	
TELUS_TC34	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 Repeat the same test using SIP REFER	Pass	
TELUS_TC35	TELUS VoIP client calls IP PBX phone 1 IP PBX phone 1 performs a blind transfer to Telus mobile client  Repeat the same test using SIP REFER	Pass	
Call Transfer (Consult transfer)			
TELUS_TC36	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 Repeat the same test using SIP REFER	Pass	
TELUS_TC37	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 Repeat the same test using SIP REFER	Pass	
TELUS_TC38	TELUS VoIP client calls IP PBX phone 1 IP PBX phone 1 performs a consult transfer to Telus mobile client Repeat the same test using SIP REFER	Pass	
Call Forwarding Unconditional			
TELUS_TC39	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_TC40	Configure IP PBX phone 1 to CFU to 1-877-353-9586 TELUS VoIP client calls phone 1 to trigger the call forwarding 1. Confirm 2-way voice 2. Press 1234# to interrupt the prompt	Pass	
TELUS_TC38	TELUS VoIP client calls IP PBX phone 1 IP PBX phone 1 performs a consult transfer to Telus mobile client Repeat the same test using SIP REFER	Pass	
Voicemail			
TELUS_TC41	Repeat for both G.711 and G.729. 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	

Conference call			
TELUS_TC42	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	
TELUS_TC44	IP PBX phone 1 calls TELUS VoIP client BVOIP performs a conference call to 1-877-353-9586 Confirm audio with VoIP client and IP PBX phone	Pass	
Test with TELUS mobile			
Pre-test Procedure / Requirements			
Schedule test time with TELUS account representative. TELUS contact is required to support mobile testing. TELUS IOT contact will share mobile number at time of test.			
Basic inbound/outbound call			
TELUS_TC45	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_TC46	Repeat the test by setup the call with both G.711 and 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	
Basic inbound/outbound call with privacy			
TELUS_TC47	Call from TELUS mobile 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_TC48	Call from IP PBX phone 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	NA	
Hold and resume			
TELUS_TC49	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_TC50	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	
Call Transfer (Blind transfer)			

TELUS_TC51	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 Repeat the same test using SIP REFER	Pass	
TELUS_TC52	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 Repeat the same test using SIP REFER	Pass	
TELUS_TC53	TELUS mobile client calls IP PBX phone 1 IP PBX phone 1 performs a blind transfer to 1-877-353-9586 Confirm the prompt and interrupt the prompt with 1234# Repeat the same test using SIP REFER	Pass	
TELUS_TC52	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 Repeat the same test using SIP REFER	Pass	
Call Transfer (Consult transfer)			
TELUS_TC54	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 Repeat the same test using SIP REFER	Pass	
TELUS_TC55	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 Repeat the same test using SIP REFER	Pass	
TELUS_TC56	TELUS mobile client calls IP PBX phone 1 IP PBX phone 1 performs a consult transfer to 1-877-353-9586 Confirm the prompt and interrupt the prompt with 1234# Repeat the same test using SIP REFER	Pass	
TELUS_TC52	TELUS mobile client calls IP PBX phone 1 IP PBX phone 1 performs a blind transfer to another Telys Mobile client Confirm 2-way voice after the transfer Repeat the same test using SIP REFER	Pass	
Call Forwarding Don't Answer			
TELUS_TC58	Configure a Mobile Phone to Forward calls to a PSTN when Dont Answer. 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	
Call Forwarding Unconditional			

TELUS_TC59	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	
TELUS_TC60	Configure IP PBX phone 1 to CFU to 1-877-353-9586 TELUS mobile client calls phone 1 to trigger the call forwarding 1. Confirm audio prompt 2. Press 1234# to interrupt the prompt	Fail	audio for 2 second and then to noway audio
Voicemail			
TELUS_TC61	Repeat the test by setup the call with G.711 and G.729. 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	
Conference call			
TELUS_TC62	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	
DTMF			
TELUS_TC65	From PBX dial 4036929600 ( 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_TC66	From PBX dial 1-877-353-9586 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 RFC2833 by programming PBX endpoint	Pass	
Automatic Blocking			
TELUS_TC72	Automatic Blocking Feature to be setup for the SIP PBX in the switch. Call from SIP PBX to a Bell Land Line Number. 1. Confirm 2-way voice 2. Confirm the proper calling number (IPTR2 DID or Alternate Number from SIP PBX) is not shown 3. Confirm that SIP PBX is not sending out Name in the call.	NA	

## 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

```
certificate-record
  name                ESBCCert1
  country              US
  state                MA
  locality             Burlington
  organization         Engineering
  unit
  common-name         lync-acme-
sbc.teluscpslynclab.net
  key-size            2048
  alternate-name
  trusted              enabled
  key-usage-list      digitalSignature
                      keyEncipherment
  extended-key-usage-list
  options
certificate-record
  name                MediationRoot
  country              US
  state                MA
  locality             Burlington
  organization         Engineering
  unit
  common-name         teluscpslynclab-DC1-LYNCLAB-CA-
1
  key-size            2048
  alternate-name
  trusted              enabled
  key-usage-list      digitalSignature
                      keyEncipherment
  extended-key-usage-list
  options
local-policy
  from-address        *
  to-address          *
  source-realm        inside
  description
  activate-time
  deactivate-time
  state                enabled
  policy-priority     none
  policy-attribute
  next-hop            10.27.56.7
```

```

realm outside
action none
terminate-recursion disabled
carrier
start-time 0000
end-time 2400
days-of-week U-S
cost 0
state enabled
app-protocol
methods
media-profiles
lookup single
next-key
eloc-str-lkup disabled
eloc-str-match

local-policy
  from-address *
  to-address *
  source-realm outside
  description
  activate-time
  deactivate-time
  state enabled
  policy-priority none
  policy-attribute
    next-hop SAG:med-grp-1
    realm inside
    action replace-uri
    terminate-recursion disabled
    carrier
    start-time 0000
    end-time 2400
    days-of-week U-S
    cost 0
    state enabled
    app-protocol SIP
    methods
    media-profiles
    lookup single
    next-key
    eloc-str-lkup disabled
    eloc-str-match

local-policy
  from-address *
  to-address fe0101.teluscpslynclab.net
  source-realm outside
  description
  activate-time
  deactivate-time
  state enabled
  policy-priority none
  policy-attribute
    next-hop
fe0101.teluscpslynclab.net
  realm inside

```



```

        action                replace-uri
        terminate-recursion   disabled
        carrier
        start-time            0000
        end-time               2400
        days-of-week          U-S
        cost                   0
        state                  enabled
        app-protocol
        methods
        media-profiles
        lookup                 single
        next-key
        eloc-str-lkup         disabled
        eloc-str-match

local-policy
    from-address             *
    to-address               fe0102.teluscpslynclab.net
    source-realm             outside
    description
    activate-time
    deactivate-time
    state                    enabled
    policy-priority         none
    policy-attribute
        next-hop
fe0102.teluscpslynclab.net
    realm                   inside
    action                  replace-uri
    terminate-recursion     disabled
    carrier
    start-time              0000
    end-time                 2400
    days-of-week            U-S
    cost                     0
    state                    enabled
    app-protocol
    methods
    media-profiles
    lookup                   single
    next-key
    eloc-str-lkup           disabled
    eloc-str-match

local-policy
    from-address             *
    to-address               fe0103.teluscpslynclab.net
    source-realm             outside
    description
    activate-time
    deactivate-time
    state                    enabled
    policy-priority         none
    policy-attribute
        next-hop
fe0103.teluscpslynclab.net
    realm                   inside
    action                  replace-uri

```

terminate-recursion	disabled
carrier	
start-time	0000
end-time	2400
days-of-week	U-S
cost	0
state	enabled
app-protocol	
methods	
media-profiles	
lookup	single
next-key	
eloc-str-lkup	disabled
eloc-str-match	
media-manager	
state	enabled
latching	enabled
flow-time-limit	86400
initial-guard-timer	300
subsq-guard-timer	300
tcp-flow-time-limit	86400
tcp-initial-guard-timer	300
tcp-subsq-guard-timer	300
tcp-number-of-ports-per-flow	2
hnt-rtcp	disabled
algd-log-level	NOTICE
mbcd-log-level	NOTICE
options	
red-flow-port	1985
red-mgcp-port	1986
red-max-trans	10000
red-sync-start-time	5000
red-sync-comp-time	1000
media-policing	enabled
max-untrusted-packet-rate	50000
max-trusted-packet-rate	50000
max-arp-packet-rate	1000
tolerance-window	30
trap-on-demote-to-deny	disabled
trap-on-demote-to-untrusted	disabled
syslog-on-demote-to-deny	disabled
syslog-on-demote-to-untrusted	disabled
rtcp-rate-limit	0
anonymous-sdp	disabled
rfc2833-timestamp	disabled
default-2833-duration	100
rfc2833-end-pkts-only-for-non-sig	enabled
translate-non-rfc2833-event	disabled
media-supervision-traps	disabled
dnalg-server-failover	disabled
syslog-on-call-reject	disabled
media-sec-policy	
name	rtponly
pass-through	disabled
options	
inbound	

```

        profile
        mode                               rtp
        protocol                           none
    outbound
        profile
        mode                               rtp
        protocol                           none
media-sec-policy
    name                                  sdespolicy
    pass-through                          disabled
    options
    inbound
        profile                            sdes1
        mode                               srtp
        protocol                           sdes
    outbound
        profile                            sdes1
        mode                               srtp
        protocol                           sdes
network-interface
    name                                  s0p0
    sub-port-id                            0
    description                            Outside/Untrusted
    hostname
    ip-address                             172.16.153.34
    pri-utility-addr                       172.16.153.2
    sec-utility-addr                       172.16.153.3
    netmask                                 255.255.255.0
    gateway                                 172.16.153.1
    sec-gateway
    gw-heartbeat
        state                              disabled
        heartbeat                           0
        retry-count                         0
        retry-timeout                       1
        health-score                        0
    dns-ip-primary
    dns-ip-backup1
    dns-ip-backup2
    dns-domain
    dns-timeout                            11
    signaling-mtu                           0
    hip-ip-list
    ftp-address
    icmp-address
    snmp-address
    telnet-address
    ssh-address
network-interface
    name                                  s1p0
    sub-port-id                            0
    description                            Inside/Trusted
    hostname
    ip-address                             172.16.154.35
    pri-utility-addr                       172.16.154.2
    sec-utility-addr                       172.16.154.3

```

```

netmask                255.255.255.0
gateway                172.16.154.1
sec-gateway
gw-heartbeat
    state                disabled
    heartbeat            0
    retry-count          0
    retry-timeout        1
    health-score         0
dns-ip-primary
dns-ip-backup1
dns-ip-backup2
dns-domain
dns-timeout            11
signaling-mtu          0
hip-ip-list
ftp-address
icmp-address
snmp-address
telnet-address
ssh-address
network-interface
    name                  wancom1
    sub-port-id           0
    description           HA HEARTBEAT1
    hostname
    ip-address
    pri-utility-addr      169.254.1.1
    sec-utility-addr      169.254.1.2
    netmask                255.255.255.252
    gateway
    sec-gateway
    gw-heartbeat
        state                disabled
        heartbeat            0
        retry-count          0
        retry-timeout        1
        health-score         0
    dns-ip-primary
    dns-ip-backup1
    dns-ip-backup2
    dns-domain
    dns-timeout            11
    hip-ip-list
    ftp-address
    icmp-address
    snmp-address
    telnet-address
    ssh-address
network-interface
    name                  wancom2
    sub-port-id           0
    description           HA HEARTBEAT2
    hostname
    ip-address
    pri-utility-addr      169.254.2.1

```

```

sec-utility-addr          169.254.2.2
netmask                   255.255.255.252
gateway
sec-gateway
gw-heartbeat
    state                 disabled
    heartbeat             0
    retry-count           0
    retry-timeout         1
    health-score          0
dns-ip-primary
dns-ip-backup1
dns-ip-backup2
dns-domain
dns-timeout               11
hip-ip-list
ftp-address
icmp-address
snmp-address
telnet-address
ssh-address
server
phy-interface
    name                  s0p0
    operation-type        Media
    port                  0
    slot                  0
    virtual-mac
    admin-state           enabled
    auto-negotiation      enabled
    duplex-mode           FULL
    speed                 100
    wancom-health-score   50
    overload-protection   disabled
phy-interface
    name                  s1p0
    operation-type        Media
    port                  0
    slot                  1
    virtual-mac
    admin-state           enabled
    auto-negotiation      enabled
    duplex-mode           FULL
    speed                 100
    wancom-health-score   50
    overload-protection   disabled
realm-config
    identifier            inside
    description
    addr-prefix           0.0.0.0
    network-interfaces    s0p0:0
    mm-in-realm           disabled
    mm-in-network         enabled
    mm-same-ip            enabled
    mm-in-system          enabled
    bw-cac-non-mm         disabled

```

msm-release	disabled
qos-enable	disabled
max-bandwidth	0
fallback-bandwidth	0
max-priority-bandwidth	0
max-latency	0
max-jitter	0
max-packet-loss	0
observ-window-size	0
parent-realm	
dns-realm	
media-policy	
media-sec-policy	sdespolicy
srtp-msm-passthrough	disabled
class-profile	
in-translationid	
out-translationid	
in-manipulationid	
out-manipulationid	
average-rate-limit	0
access-control-trust-level	none
invalid-signal-threshold	0
maximum-signal-threshold	0
untrusted-signal-threshold	0
nat-trust-threshold	0
max-endpoints-per-nat	0
nat-invalid-message-threshold	0
wait-time-for-invalid-register	0
deny-period	30
cac-failure-threshold	0
untrust-cac-failure-threshold	0
ext-policy-svr	
diam-e2-address-realm	
subscription-id-type	END USER NONE
symmetric-latching	disabled
pai-strip	disabled
trunk-context	
device-id	
early-media-allow	
enforcement-profile	
additional-prefixes	
restricted-latching	none
restriction-mask	32
user-cac-mode	none
user-cac-bandwidth	0
user-cac-sessions	0
icmp-detect-multiplier	0
icmp-advertisement-interval	0
icmp-target-ip	
monthly-minutes	0
options	
spl-options	
accounting-enable	enabled
net-management-control	disabled
delay-media-update	disabled
refer-call-transfer	disabled

refer-notify-provisional	none
dyn-refer-term	disabled
codec-policy	
codec-manip-in-realm	disabled
codec-manip-in-network	enabled
rtcp-policy	
constraint-name	
session-recording-server	
session-recording-required	disabled
manipulation-string	
manipulation-pattern	
stun-enable	disabled
stun-server-ip	0.0.0.0
stun-server-port	3478
stun-changed-ip	0.0.0.0
stun-changed-port	3479
sip-profile	
sip-isup-profile	
match-media-profiles	
qos-constraint	
block-rtcp	disabled
hide-egress-media-update	disabled
tcp-media-profile	
monitoring-filters	
node-functionality	
default-location-string	
alt-family-realm	
pref-addr-type	none
realm-config	
identifier	outside
description	
addr-prefix	0.0.0.0
network-interfaces	s0p1:0
mm-in-realm	disabled
mm-in-network	enabled
mm-same-ip	enabled
mm-in-system	enabled
bw-cac-non-mm	disabled
msm-release	disabled
qos-enable	disabled
max-bandwidth	0
fallback-bandwidth	0
max-priority-bandwidth	0
max-latency	0
max-jitter	0
max-packet-loss	0
observ-window-size	0
parent-realm	
dns-realm	
media-policy	
media-sec-policy	rtponly
srtp-msm-passthrough	disabled
class-profile	
in-translationid	
out-translationid	
in-manipulationid	

out-manipulationid	
average-rate-limit	0
access-control-trust-level	none
invalid-signal-threshold	0
maximum-signal-threshold	0
untrusted-signal-threshold	0
nat-trust-threshold	0
max-endpoints-per-nat	0
nat-invalid-message-threshold	0
wait-time-for-invalid-register	0
deny-period	30
cac-failure-threshold	0
untrust-cac-failure-threshold	0
ext-policy-svr	
diam-e2-address-realm	
subscription-id-type	END USER NONE
symmetric-latching	disabled
pai-strip	disabled
trunk-context	
device-id	
early-media-allow	
enforcement-profile	
additional-prefixes	
restricted-latching	none
restriction-mask	32
user-cac-mode	none
user-cac-bandwidth	0
user-cac-sessions	0
icmp-detect-multiplier	0
icmp-advertisement-interval	0
icmp-target-ip	
monthly-minutes	0
options	
spl-options	
accounting-enable	enabled
net-management-control	disabled
delay-media-update	disabled
refer-call-transfer	disabled
refer-notify-provisional	none
dyn-refer-term	disabled
codec-policy	
codec-manip-in-realm	disabled
codec-manip-in-network	enabled
rtcp-policy	
constraint-name	
session-recording-server	
session-recording-required	disabled
manipulation-string	
manipulation-pattern	
stun-enable	disabled
stun-server-ip	0.0.0.0
stun-server-port	3478
stun-changed-ip	0.0.0.0
stun-changed-port	3479
sip-profile	
sip-isup-profile	



```

match-media-profiles
qos-constraint
block-rtcp                               disabled
hide-egress-media-update                 disabled
tcp-media-profile
monitoring-filters
node-functionality
default-location-string
alt-family-realm
pref-addr-type                           none
last-modified-by                         admin@192.168.20.104
last-modified-date                       2015-07-24 02:16:49
sdes-profile
  name                                    sdes1
  crypto-list                             AES CM 128 HMAC SHA1 80
                                           AES CM 128 HMAC SHA1 32
  srtp-auth                               enabled
  srtp-encrypt                            enabled
  srtcp-encrypt                           enabled
  mki                                      disabled
  egress-offer-format                     simultaneous-best-effort
  use-ingress-session-params
  options
  key
  salt
session-agent
  hostname                                10.27.56.7
  ip-address                              10.27.56.7
  port                                    5060
  state                                   enabled
  app-protocol                            SIP
  app-type
  transport-method                        UDP
  realm-id                                outside
  egress-realm-id
  description
  carriers
  allow-next-hop-lp                       enabled
  constraints                             disabled
  max-sessions                            0
  max-inbound-sessions                    0
  max-outbound-sessions                    0
  max-burst-rate                          0
  max-inbound-burst-rate                  0
  max-outbound-burst-rate                 0
  max-sustain-rate                        0
  max-inbound-sustain-rate                0
  max-outbound-sustain-rate               0
  min-seizures                            5
  min-asr                                 0
  time-to-resume                          0
  ttr-no-response                         0
  in-service-period                       0
  burst-rate-window                       0
  sustain-rate-window                     0
  req-uri-carrier-mode                     None

```

```

proxy-mode
redirect-action
loose-routing                enabled
send-media-session          enabled
response-map
ping-method                  OPTIONS
ping-interval                90
ping-send-mode               keep-alive
ping-all-addresses          disabled
ping-in-service-response-codes
out-service-response-codes
load-balance-dns-query      hunt
options
spl-options
media-profiles
in-translationid
out-translationid
trust-me                      disabled
request-uri-headers
stop-recurse
local-response-map
ping-to-user-part
ping-from-user-part
in-manipulationid
out-manipulationid          To Telus
manipulation-string
manipulation-pattern
p-asserted-id
trunk-group
max-register-sustain-rate    0
early-media-allow
invalidate-registrations     disabled
rfc2833-mode                 none
rfc2833-payload              0
codec-policy
enforcement-profile
refer-call-transfer          disabled
refer-notify-provisional    none
reuse-connections           NONE
tcp-keepalive                none
tcp-reconn-interval          0
max-register-burst-rate      0
register-burst-window         0
sip-profile
sip-isup-profile
kpml-interworking            inherit
monitoring-filters
session-recording-server
session-recording-required   disabled
session-agent
  hostname                    fe0101.teluscpslynclab.net
  ip-address                   172.16.149.38
  port                          5067
  state                         enabled
  app-protocol                  SIP
  app-type

```

transport-method	StaticTLS
realm-id	inside
egress-realm-id	
description	
carriers	
allow-next-hop-lp	enabled
constraints	disabled
max-sessions	0
max-inbound-sessions	0
max-outbound-sessions	0
max-burst-rate	0
max-inbound-burst-rate	0
max-outbound-burst-rate	0
max-sustain-rate	0
max-inbound-sustain-rate	0
max-outbound-sustain-rate	0
min-seizures	5
min-asr	0
time-to-resume	0
ttr-no-response	0
in-service-period	0
burst-rate-window	0
sustain-rate-window	0
req-uri-carrier-mode	None
proxy-mode	
redirect-action	
loose-routing	enabled
send-media-session	enabled
response-map	
ping-method	
ping-interval	60
ping-send-mode	keep-alive
ping-all-addresses	disabled
ping-in-service-response-codes	
out-service-response-codes	
load-balance-dns-query	hunt
options	
spl-options	
media-profiles	
in-translationid	
out-translationid	
trust-me	disabled
request-uri-headers	
stop-recurse	
local-response-map	
ping-to-user-part	
ping-from-user-part	
in-manipulationid	
out-manipulationid	
manipulation-string	
manipulation-pattern	
p-asserted-id	
trunk-group	
max-register-sustain-rate	0
early-media-allow	
invalidate-registrations	disabled

rfc2833-mode	none
rfc2833-payload	0
codec-policy	
enforcement-profile	
refer-call-transfer	enabled
refer-notify-provisional	none
reuse-connections	NONE
tcp-keepalive	none
tcp-reconn-interval	0
max-register-burst-rate	0
register-burst-window	0
sip-profile	
sip-isup-profile	
kpml-interworking	inherit
monitoring-filters	
session-recording-server	
session-recording-required	disabled
session-agent	
hostname	fe0102.teluscpslynclab.net
ip-address	172.16.149.39
port	5067
state	enabled
app-protocol	SIP
app-type	
transport-method	StaticTLS
realm-id	inside
egress-realm-id	
description	
carriers	
allow-next-hop-lp	enabled
constraints	disabled
max-sessions	0
max-inbound-sessions	0
max-outbound-sessions	0
max-burst-rate	0
max-inbound-burst-rate	0
max-outbound-burst-rate	0
max-sustain-rate	0
max-inbound-sustain-rate	0
max-outbound-sustain-rate	0
min-seizures	5
min-asr	0
time-to-resume	0
ttr-no-response	0
in-service-period	0
burst-rate-window	0
sustain-rate-window	0
req-uri-carrier-mode	None
proxy-mode	
redirect-action	
loose-routing	enabled
send-media-session	enabled
response-map	
ping-method	
ping-interval	60
ping-send-mode	keep-alive

ping-all-addresses	disabled
ping-in-service-response-codes	
out-service-response-codes	
load-balance-dns-query	hunt
options	
spl-options	
media-profiles	
in-translationid	
out-translationid	
trust-me	disabled
request-uri-headers	
stop-recurse	
local-response-map	
ping-to-user-part	
ping-from-user-part	
in-manipulationid	
out-manipulationid	
manipulation-string	
manipulation-pattern	
p-asserted-id	
trunk-group	
max-register-sustain-rate	0
early-media-allow	
invalidate-registrations	disabled
rfc2833-mode	none
rfc2833-payload	0
codec-policy	
enforcement-profile	
refer-call-transfer	enabled
refer-notify-provisional	none
reuse-connections	NONE
tcp-keepalive	none
tcp-reconn-interval	0
max-register-burst-rate	0
register-burst-window	0
sip-profile	
sip-isup-profile	
kpml-interworking	inherit
monitoring-filters	
session-recording-server	
session-recording-required	disabled
session-agent	
hostname	fe0103.teluscpslynclab.net
ip-address	172.16.149.40
port	5067
state	enabled
app-protocol	SIP
app-type	
transport-method	StaticTLS
realm-id	inside
egress-realm-id	
description	
carriers	
allow-next-hop-lp	enabled
constraints	disabled
max-sessions	0

max-inbound-sessions	0
max-outbound-sessions	0
max-burst-rate	0
max-inbound-burst-rate	0
max-outbound-burst-rate	0
max-sustain-rate	0
max-inbound-sustain-rate	0
max-outbound-sustain-rate	0
min-seizures	5
min-asr	0
time-to-resume	0
ttr-no-response	0
in-service-period	0
burst-rate-window	0
sustain-rate-window	0
req-uri-carrier-mode	None
proxy-mode	
redirect-action	
loose-routing	enabled
send-media-session	enabled
response-map	
ping-method	
ping-interval	60
ping-send-mode	keep-alive
ping-all-addresses	disabled
ping-in-service-response-codes	
out-service-response-codes	
load-balance-dns-query	hunt
options	
spl-options	
media-profiles	
in-translationid	
out-translationid	
trust-me	disabled
request-uri-headers	
stop-recurse	
local-response-map	
ping-to-user-part	
ping-from-user-part	
in-manipulationid	
out-manipulationid	
manipulation-string	
manipulation-pattern	
p-asserted-id	
trunk-group	
max-register-sustain-rate	0
early-media-allow	
invalidate-registrations	disabled
rfc2833-mode	none
rfc2833-payload	0
codec-policy	
enforcement-profile	
refer-call-transfer	enabled
refer-notify-provisional	none
reuse-connections	NONE
tcp-keepalive	none

```

tcp-reconn-interval          0
max-register-burst-rate     0
register-burst-window       0
sip-profile
sip-isup-profile
kpml-interworking           inherit
monitoring-filters
session-recording-server
session-recording-required  disabled
session-group
  group-name                 med-grp-1
  description                Lync Mediation server group
  state                      enabled
  app-protocol               SIP
  strategy                   Hunt
  dest                       fe0101.teluscpslyncclab.net
                             fe0102.teluscpslyncclab.net
                             fe0103.teluscpslyncclab.net

  trunk-group
  sag-recursion              disabled
  stop-sag-recurse           401,407
  last-modified-by           admin@172.21.0.93
  last-modified-date         2015-05-14 19:51:34
sip-config
  state                      enabled
  operation-mode              dialog
  dialog-transparency         enabled
  home-realm-id               inside
  egress-realm-id
  auto-realm-id
  nat-mode                    None
  registrar-domain            *
  registrar-host               *
  registrar-port               0
  register-service-route       always
  init-timer                   500
  max-timer                    4000
  trans-expire                 32
  initial-inv-trans-expire     0
  invite-expire                180
  inactive-dynamic-conn       32
  enforcement-profile
  pac-method
  pac-interval                 10
  pac-strategy                 PropDist
  pac-load-weight              1
  pac-session-weight           1
  pac-route-weight             1
  pac-callid-lifetime          600
  pac-user-lifetime            3600
  red-sip-port                 1988
  red-max-trans                10000
  red-sync-start-time          5000
  red-sync-comp-time           1000
  options                      max-udp-length=0
  add-reason-header            disabled

```

```

sip-message-len                4096
enum-sag-match                 disabled
extra-method-stats             disabled
extra-enum-stats               disabled
rph-feature                    disabled
nsep-user-sessions-rate        0
nsep-sa-sessions-rate          0
registration-cache-limit       0
register-use-to-for-lp          disabled
refer-src-routing              disabled
add-ucid-header                disabled
proxy-sub-events               disabled
allow-pani-for-trusted-only    disabled
atcf-stn-sr                    disabled
atcf-psi-dn                    disabled
atcf-route-to-sccas            disabled
eatf-stn-sr                    disabled
pass-gruu-contact              disabled
sag-lookup-on-redirect         disabled
set-disconnect-time-on-bye     disabled
msrp-delayed-bye-timer         15
transcoding-realm              disabled
transcoding-agents             disabled
create-dynamic-sa              disabled
node-functionality             P-CSCF
last-modified-by               admin@172.21.0.93
last-modified-date             2015-05-29 20:46:50
sip-interface
state                           enabled
realm-id                        inside
description
sip-port
    address                      172.16.153.34
    port                          5066
    transport-protocol            TLS
    tls-profile                   Core
    allow-anonymous               agents-only
    multi-home-addr               disabled
    ims-aka-profile               disabled
carriers
trans-expire                     0
initial-inv-trans-expire         0
invite-expire                    0
max-redirect-contacts           0
proxy-mode                       disabled
redirect-action                  disabled
contact-mode                     none
nat-traversal                    none
nat-interval                     30
tcp-nat-interval                 90
registration-caching             disabled
min-reg-expire                   300
registration-interval            3600
route-to-registrar               disabled
secured-network                  disabled
teluri-scheme                    disabled

```



uri-fqdn-domain	
options	
spl-options	
trust-mode	all
max-nat-interval	3600
nat-int-increment	10
nat-test-increment	30
sip-dynamic-hnt	disabled
stop-recurse	401,407
port-map-start	0
port-map-end	0
in-manipulationid	
out-manipulationid	
sip-ims-feature	disabled
sip-atcf-feature	disabled
subscribe-reg-event	disabled
operator-identifier	
anonymous-priority	none
max-incoming-conns	0
per-src-ip-max-incoming-conns	0
inactive-conn-timeout	0
untrusted-conn-timeout	0
network-id	
ext-policy-server	
ldap-policy-server	
default-location-string	
term-tgrp-mode	none
charging-vector-mode	pass
charging-function-address-mode	pass
ccf-address	
ecf-address	
implicit-service-route	disabled
rfc2833-payload	101
rfc2833-mode	transparent
constraint-name	
response-map	
local-response-map	
ims-aka-feature	disabled
enforcement-profile	
route-unauthorized-calls	
tcp-keepalive	none
add-sdp-invite	disabled
add-sdp-profiles	
manipulation-string	
manipulation-pattern	
sip-profile	
sip-isup-profile	
tcp-conn-dereg	0
tunnel-name	
register-keep-alive	none
kpml-interworking	disabled
msrp-delay-egress-bye	disabled
send-380-response	
pcscf-restoration	
session-timer-profile	
session-recording-server	

```

session-recording-required      disabled
service-tag
sip-interface
state                            enabled
realm-id                          outside
description
sip-port
    address                       172.16.154.35
    port                           5060
    transport-protocol             UDP
    tls-profile
    allow-anonymous                all
    multi-home-addr
    ims-aka-profile
sip-port
    address                       172.16.154.35
    port                           5061
    transport-protocol             TLS
    tls-profile                     Outside
    allow-anonymous                agents-only
    multi-home-addr
    ims-aka-profile
carriers
trans-expire                       0
initial-inv-trans-expire           0
invite-expire                       0
max-redirect-contacts              0
proxy-mode
redirect-action
contact-mode                        none
nat-traversal                       none
nat-interval                         30
tcp-nat-interval                    90
registration-caching                disabled
min-reg-expire                       300
registration-interval               3600
route-to-registrar                  disabled
secured-network                      disabled
teluri-scheme                       disabled
uri-fqdn-domain
options
spl-options
trust-mode                          all
max-nat-interval                    3600
nat-int-increment                   10
nat-test-increment                  30
sip-dynamic-hnt                     disabled
stop-recurse                         401,407
port-map-start                       0
port-map-end                         0
in-manipulationid
out-manipulationid
sip-ims-feature                      disabled
sip-atcf-feature                     disabled
subscribe-reg-event                  disabled
operator-identifier

```

anonymous-priority	none
max-incoming-conns	0
per-src-ip-max-incoming-conns	0
inactive-conn-timeout	0
untrusted-conn-timeout	0
network-id	
ext-policy-server	
ldap-policy-server	
default-location-string	
term-tgrp-mode	none
charging-vector-mode	pass
charging-function-address-mode	pass
ccf-address	
ecf-address	
implicit-service-route	disabled
rfc2833-payload	101
rfc2833-mode	transparent
constraint-name	
response-map	
local-response-map	
ims-aka-feature	disabled
enforcement-profile	
route-unauthorized-calls	
tcp-keepalive	none
add-sdp-invite	disabled
add-sdp-profiles	
manipulation-string	
manipulation-pattern	
sip-profile	
sip-isup-profile	
tcp-conn-dereg	0
tunnel-name	
register-keep-alive	none
kpml-interworking	disabled
msrp-delay-egress-bye	disabled
send-380-response	
pcscf-restoration	
session-timer-profile	
session-recording-server	
session-recording-required	disabled
service-tag	
sip-manipulation	
name	To Telus
description	
split-headers	
join-headers	
header-rule	
name	save PAI
header-name	P-Asserted-Identity
action	store
comparison-type	case-sensitive
msg-type	any
methods	
match-value	
new-value	
header-rule	

```

name Updt PAI
header-name P-Asserted-Identity
action add
comparison-type boolean
msg-type any
methods INVITE
match-value !$save PAI
new-value
<sip:2223334444@ipinet4.com;user=phone>
header-rule
name Updt RURI
header-name request-uri
action manipulate
comparison-type case-sensitive
msg-type any
methods
match-value
new-value
element-rule
name Updt URI Host
parameter-name
type uri-host
action replace
match-val-type any
comparison-type case-sensitive
match-value
new-value ipinet4.com
header-rule
name Updt To
header-name To
action manipulate
comparison-type case-sensitive
msg-type any
methods
match-value
new-value
element-rule
name UPdt URI host
parameter-name
type uri-host
action replace
match-val-type any
comparison-type case-sensitive
match-value
new-value ipinet4.com
header-rule
name Updt From
header-name From
action manipulate
comparison-type case-sensitive
msg-type any
methods
match-value
new-value
element-rule
name
Updt URI host

```

```

parameter-name
type uri-host
action replace
match-val-type any
comparison-type case-sensitive
match-value
new-value ipinet4.com

header-rule
name Updt Contact
header-name Contact
action manipulate
comparison-type case-sensitive
msg-type any
methods
match-value
new-value
element-rule
name Updt URI Host
parameter-name
type uri-host
action replace
match-val-type any
comparison-type case-sensitive
match-value
new-value $LOCAL IP

element-rule
name Del MSOpaque
parameter-name ms-opaque
type uri-param
action delete-element
match-val-type any
comparison-type case-sensitive
match-value
new-value

header-rule
name CheckPrivacy
header-name request-uri
action store
comparison-type case-sensitive
msg-type any
methods INVITE
match-value
new-value
element-rule
name CheckStar67
parameter-name
type uri-user
action store
match-val-type any
comparison-type pattern-rule
match-value \*67\+
new-value

header-rule
name AddPrivacyHdr
header-name Privacy
action add

```

```

comparison-type          boolean
msg-type                 request
methods                  INVITE
match-value              $CheckPrivacy.$CheckStar67
new-value                id
header-rule
  name                   updateRURI
  header-name            request-uri
  action                 manipulate
  comparison-type        pattern-rule
  msg-type               request
  methods                INVITE
  match-value            new-value
  element-rule
    name                 updateRURIUser
    parameter-name
    type                 uri-user
    action               replace
    match-val-type       any
    comparison-type      pattern-rule
    match-value          \*67(.*)
    new-value            $1
header-rule
  name                   updateTO
  header-name            To
  action                 manipulate
  comparison-type        pattern-rule
  msg-type               request
  methods                INVITE
  match-value            new-value
  element-rule
    name                 updateTOUsr
    parameter-name
    type                 uri-user
    action               replace
    match-val-type       any
    comparison-type      pattern-rule
    match-value          \*67(.*)
    new-value            $1
header-rule
  name                   StoreFromTag
  header-name            From
  action                 store
  comparison-type        case-sensitive
  msg-type               request
  methods                INVITE
  match-value            new-value
  element-rule
    name                 storeTag
    parameter-name       tag
    type                 header-param
    action               store

```

```

match-val-type any
comparison-type case-sensitive
match-value
new-value

header-rule
  name ChgFromPrivacy
  header-name From
  action manipulate
  comparison-type boolean
  msg-type request
  methods INVITE
  match-value $CheckPrivacy.$CheckStar67
  new-value "\"Anonymous\"
<sip:anonymous@anonymous.invalid>; tag="+$StoreFromTag.$storeTag.$0

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

steering-pool
  ip-address 172.16.153.34
  start-port 40000
  end-port 60000
  realm-id inside
  network-interface

steering-pool
  ip-address 172.16.154.35
  start-port 49152
  end-port 57500
  realm-id outside
  network-interface

system-config
  hostname
  description
  location
  mib-system-contact
  mib-system-name
  mib-system-location
  snmp-enabled enabled
  enable-snmp-auth-traps disabled
  enable-snmp-syslog-notify disabled
  enable-snmp-monitor-traps disabled
  enable-env-monitor-traps disabled
  snmp-syslog-his-table-length 1
  snmp-syslog-level WARNING
  system-log-level WARNING
  process-log-level NOTICE
  process-log-ip-address 0.0.0.0
  process-log-port 0
  collect
    sample-interval 5
    push-interval 15
    boot-state disabled
    start-time now
    end-time never

```

```

red-collect-state disabled
red-max-trans 1000
red-sync-start-time 5000
red-sync-comp-time 1000
push-success-trap-state disabled
comm-monitor
state disabled
sbc-grp-id 0
tls-profile
qos-enable enabled
call-trace disabled
internal-trace disabled
log-filter all
default-gateway 172.16.0.254
restart enabled
exceptions
telnet-timeout 0
console-timeout 0
remote-control enabled
cli-audit-trail enabled
link-redundancy-state disabled
source-routing disabled
cli-more disabled
terminal-height 24
debug-timeout 0
trap-event-lifetime 0
ids-syslog-facility -1
options
default-v6-gateway ::
ipv6-signaling-mtu 1500
ipv4-signaling-mtu 1500
cleanup-time-of-day 00:00
snmp-engine-id-suffix
snmp-agent-mode v1v2
tls-profile
name Core
end-entity-certificate ESBCCert1
trusted-ca-certificates MediationRoot
cipher-list ALL
verify-depth 10
mutual-authenticate disabled
tls-version compatibility
options
cert-status-check disabled
cert-status-profile-list
ignore-dead-responder disabled
allow-self-signed-cert disabled
last-modified-by admin@192.168.20.105
last-modified-date 2015-07-29 18:45:51
tls-profile
name Outside
end-entity-certificate ESBCCert1
trusted-ca-certificates ESBCCert1
cipher-list ALL
verify-depth 10
mutual-authenticate disabled

```



```

tls-version compatibility
options
cert-status-check disabled
cert-status-profile-list
ignore-dead-responder disabled
allow-self-signed-cert disabled
web-server-config
state enabled
inactivity-timeout 5
http-state enabled
http-port 80
https-state disabled
https-port 443
tls-profile

```

## 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, this must be explicitly configured.

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

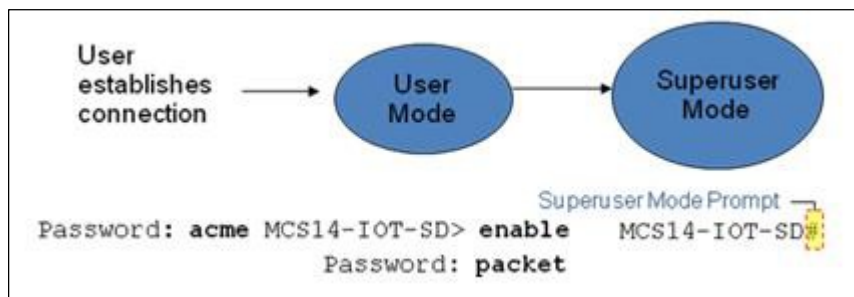
### ACLI Basics

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

When you establish a connection to the E-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.



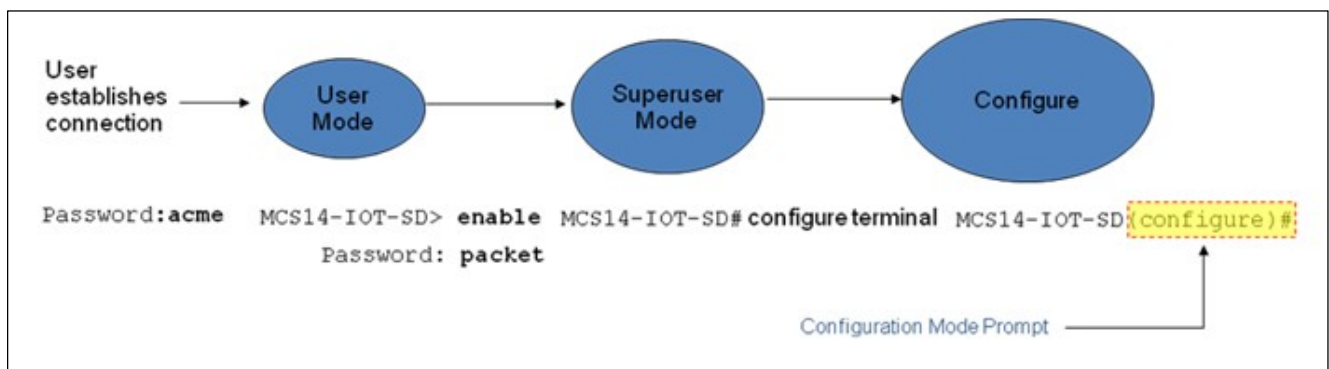
The Superuser mode allows for access to all system commands for operation, maintenance, and administration. This mode is identified by the pound sign (#) in the prompt after the target name. To enter the Superuser mode, issue the `enable` command in the User mode.



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.

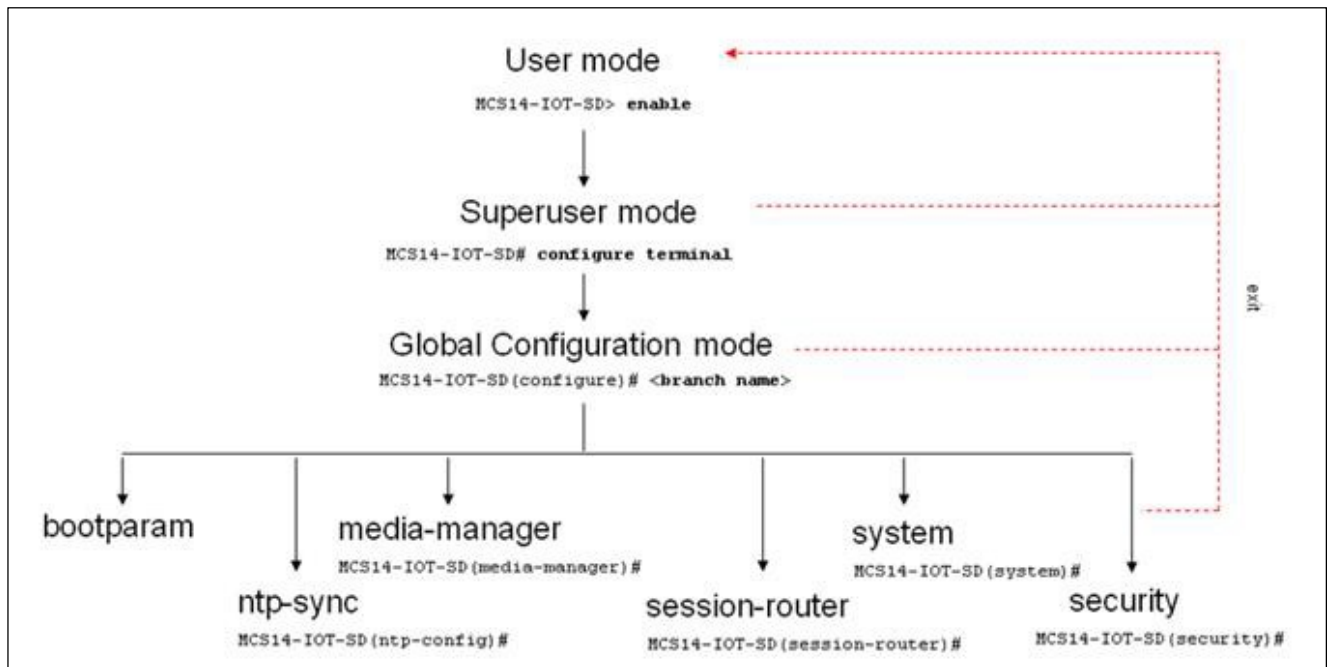
You must enter the Configuration mode to configure elements. For example, you can access the configuration branches and configuration elements for signaling and media configurations. To enter the Configuration mode, issue the `configure terminal` command in the Superuser mode.

Configuration mode is identified by the word `configure` in parenthesis followed by the pound sign (#) in the prompt after the target name, for example, `SBC1 (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 E-SBC boot parameters. Key boot parameters include:

- boot device – The global management port, usually eth0
- file name – The boot path and the image file.

- inet on ethernet – The IP address and subnet mask (in hex) of the management port of the SD.
- host inet –The IP address of external server where image file resides.
- user and ftp password – Used to boot from the external FTP server.
- gateway inet – The gateway IP address for reaching the external server, if the server is located in a different network.

```

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

```

The ntp-sync branch provides access to ntp server configuration commands for synchronizing the E-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

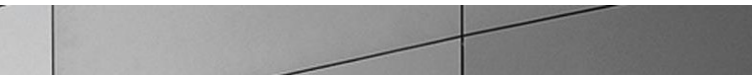
1. To create a single-instance element, you go to the appropriate level in the ACLI path and enter its parameters. There is no need to specify a unique identifier property because a single-instance element is a global element and there is only one instance of this element.
2. When creating a multiple-instance 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 E-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 E-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 E-SBC reboots, your configurations will be lost.

## Configuration Versions

At any time, three versions of the configuration can exist on the E-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 E-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 E-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 E-SBC displays a reminder on screen stating that you must use the `activate-config` command if you want the configurations to be updated.

```
SBC1 # 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'.
SBC1
```

## 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 E-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.

```
SBC1# activate-config Activate-Config
received, processing. waiting 120000 for
request to finish Request to 'ACTIVATE-
CONFIG' has Finished, Activate Complete
SBC1#
```



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