

Dinstar Session Border Controller

Training Session

Sunny Tan

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Shenzhen Dinstar Co., Ltd



Agenda 5. FAQ **♀3.** Advance features ° 1. Introduction of SBC Troubleshooting How to Configuration 64. SBC







Session Border Controller

Overview

Dinstar SBC provides rich SIP-based services such as safe network access, robust security, system interconnectivity, flexible session routing & policy management, QoS, media transcoding and media processing for small-and-medium telecommunication operators. With distributed multi-core processor, rear panel for non-blocking gigabit switching data as well as embedded Linux operating system, the session border controller delivers high capability while achieves low power dissipation.

It is able to process up to 50,000 concurrent SIP sessions and transcode 5,000 concurrent calls. Meanwhile, it supports dual control boards and dual hot-swappable power supplies, and allows encrypted sessions via TLS and SRTP. Apart form traditional codecs like G.729, G.723, G.711 and G.726, SBC3000 also supports the transcoding of iLBC, AMR and OPUS.





Session Border Controller



Security

SIP Firewall

SIP TLS/SRTP

Anti DDoS Attack

Access Control

Topology Hiding

Endpoint Authentication

SIP Intrusion Prevention

SIP Malformed Packet

Protection



Interoperability

SIP Header Manipulation

SIP Message Manipulation

B2BUA

IPv4 - IPv6

Transcoding



Session Control

Traffic Control

Registration / Call Rate

Limiting

Flexible Routing

QoS



Resiliency

High Availability

SIP Trunk Load Balancing

Alternative Routing









Session Border Controller

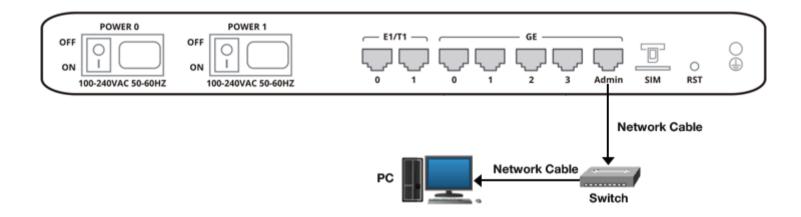








Configuration the network



The SBC Admin port use static IP 192.168.11.1/24 as default.

Before access the SBC, should add IP 192.168.11.X in your PC, like 192.168.11.20.

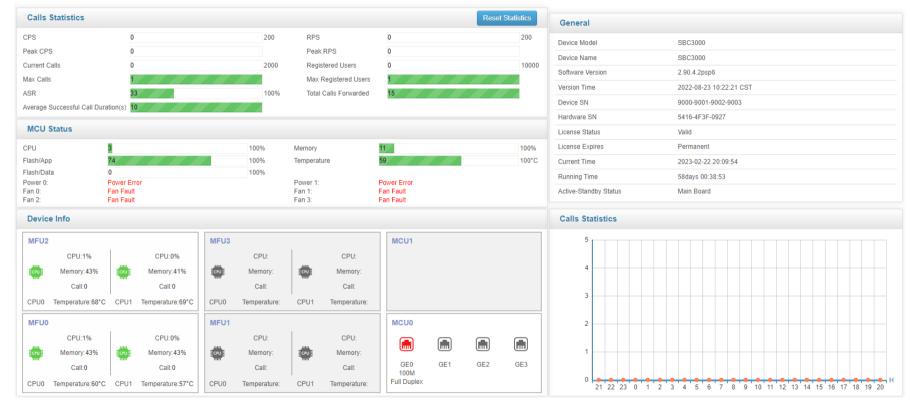
Then access the GUI of SBC through web browser.

Note:

- 1. The GUI access use HTTPS
- 2. SBC3000 used GE1



Configuration the network



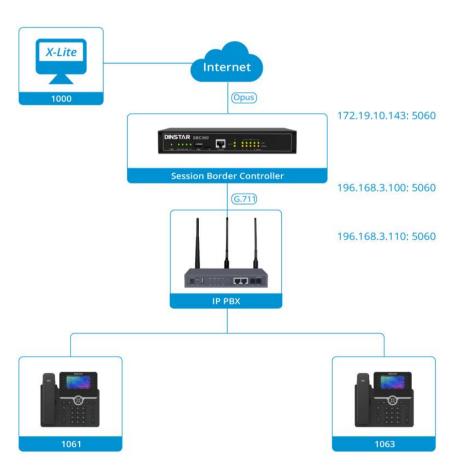


Configuration logic

- Draw the network logic graph
- Configuration the network
- Configuration the SIP trunk
- Configuration the call route
- Advance configuration (Options)



Graph of Register mode





Configuration the network

Network							
Name	Service or Management Port	MTU	Mac	IPV4 Address	Subnet Mask	IPV4 Gateway	IPV4 DNS
GE0	Service Port	1500	f8:bb:00:80:00:80	172.19.10.143	255.255.0.0	172.19.1.1	8.8.8.8/114.114.114.1 14
GE1	Service Port	1500	f8:bb:00:80:00:81	192.168.3.100	255.255.255.0	192.168.3.1	1

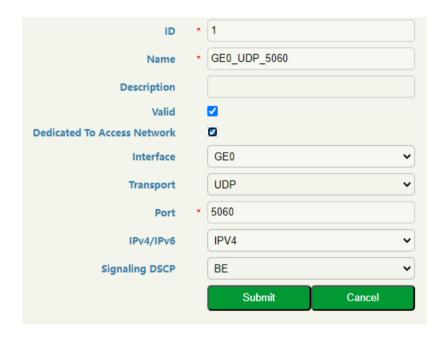
Open Page System>>>Network, set the network as graph, reboot to active. GE0 for Internet GE1 for Internal

Note:

If the SBC Ethernet port does not enough, you can use the VLAN and additional switch to extend it.



Configuration the TransPort Endpoint



Open Page System>>>TransPort Endpoint, add new rule, this one for monitor the Interface GEO and SIP port 5060.

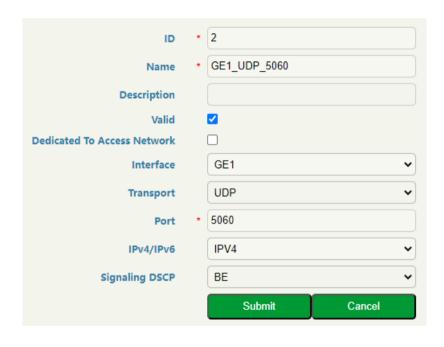
Default we use UDP, but can support to TCP or TLS.

Note:

If it will use for outside register, should enable the option "Dedicated To Access Network"



Configuration the TransPort Endpoint



Open Page System>>>TransPort Endpoint, add new rule, this one for monitor the Interface GE1 and SIP port 5060.

Default we use UDP, but can support to TCP or TLS.

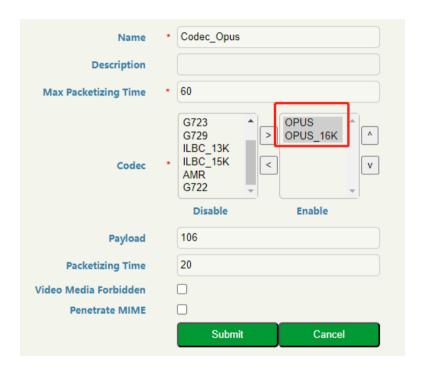


Configuration the TransPort Endpoint

TransPort Endpoint							
ID	Name	Description	Interface	Port	IPv4/IPv6	Transport	Dedicated To Access Network
1	GE0_UDP_5060		GE0	5060	ip4	udp	true
2	GE1_UDP_5060		GE1	5060	ip4	udp	false



Configuration the Codec

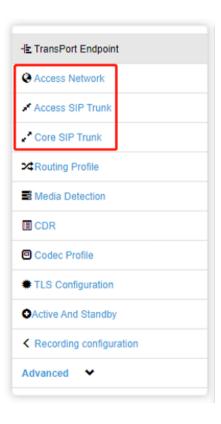


Open Page System>>>Code Profile, add a new rule.

This one for make are Codec profile, it included the OPUS only.



Configuration the SIP Trunk



The SBC have 3 types Trunks/Network.

Interface for outside:

Access Network, this one for outside register to SBC.

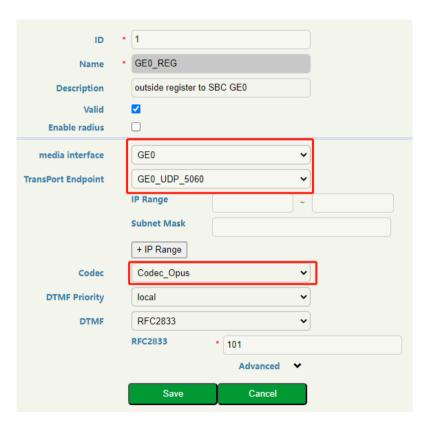
Access SIP Trunk, this one normal Trunk between SBC and Access Network.

Interface for internal.

Core SIP Trunk, this one is core Trunk between SBC and Core Network.



Configuration the SIP Trunk

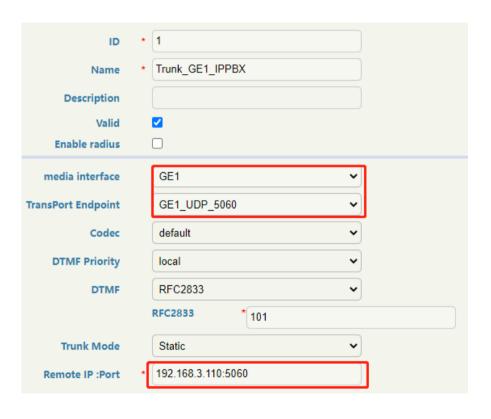


Open Page System>>>Access Network, add a new rule.

This one for make are interface for outside register to SBC, like Xlite/IPPhone.
For the Xlite/IPPhone, the SIP server IP is SBC's IP of GE0
And this trunk will use OPUS only.



Configuration the SIP Trunk



Open Page System>>>Core SIP Trunk, add a new rule.

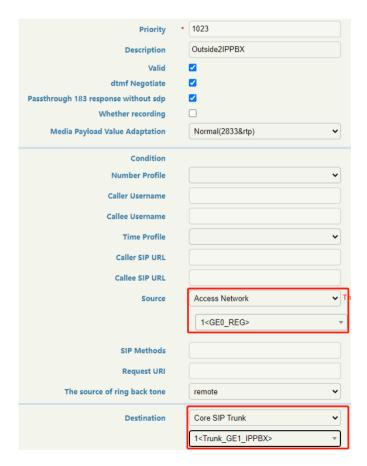
This one for make are Trunk from SBC to IPPBX.

The TransPort Endpoint choose GE1 UDP 5060

The Remote IP:port input IPPBX's IP and port



Configuration the call route



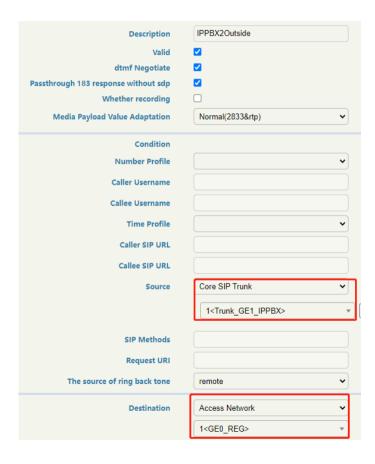
Open Page System>>>Routing
Profile>>>>CallRouting, add a new rule:
The Source choose Access Network "GE0_GE"
The Destination choose Core SIP Trunk
"Trunk_GE1_IPPBX"

This one for make are route from Outside to IPPRX.

All Register/call hit SBC's GE0 and port 5060 (172.19.10.143:5060), will out from GE1(192.168.3.100:5060) and forward to IPPBX (192.168.3.110:5060)



Configuration the call route



Open Page System>>>Routing
Profile>>>>CallRouting, add a new rule:
The Source choose Core SIP Trunk
"Trunk_GE1_IPPBX"
The Destination choose Access Network
"GE0_GE"

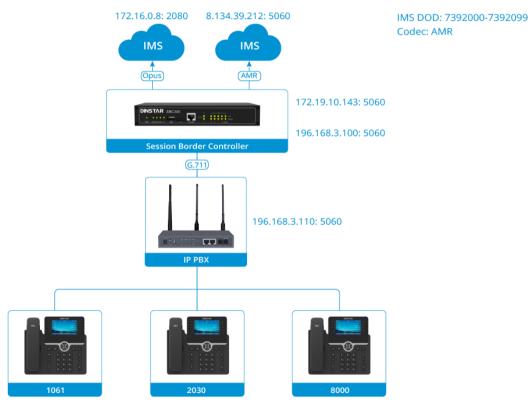
This one for make are route from IPPBX to Outside, like Xlite/IPPhone.

If a call from IPPBX (192.168.3.110:5060) hit SBC's GE1 and port 5060 (192.168.3.100:5060), it will out from GE0(172.19.10.143:5060) to outside.



Graph of Trunk mode

IMS DID: 7551000-7551099 IMS Register ID: 7551000 Password:12345678 Codec: Opus





Configuration the network

Network							
Name	Service or Management Port	MTU	Mac	IPV4 Address	Subnet Mask	IPV4 Gateway	IPV4 DNS
GE0	Service Port	1500	f8:bb:00:80:00:80	172.19.10.143	255.255.0.0	172.19.1.1	8.8.8.8/114.114.114.1 14
GE1	Service Port	1500	f8:bb:00:80:00:81	192.168.3.100	255.255.255.0	192.168.3.1	/

Open Page System>>>Network, set the network as graph, reboot to active. GE0 for External GE1 for Internal

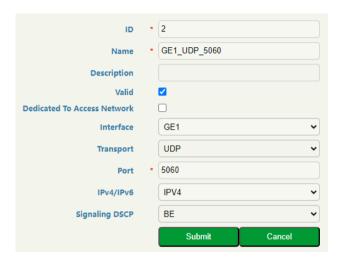
Note:

Some IMS request using VLAN, we can add the network for VLAN also.



Configuration the TransPort Endpoint





Open Page System>>>TransPort Endpoint, add two rules:

The GE0 will used for IMS

The GE1will used for IPPBX

Default using UDP, for some IMS, it request use TCP or TLS.

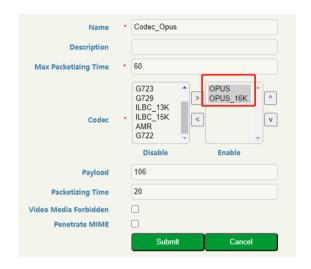


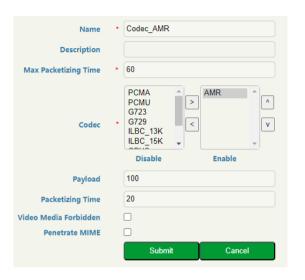
Configuration the TransPort Endpoint

TransPort Endpoint							
ID	Name	Description	Interface	Port	IPv4/IPv6	Transport	Dedicated To Access Network
1	GE0_UDP_5060		GE0	5060	ip4	udp	false
2	GE1_UDP_5060		GE1	5060	ip4	udp	false



Configuration the Codec





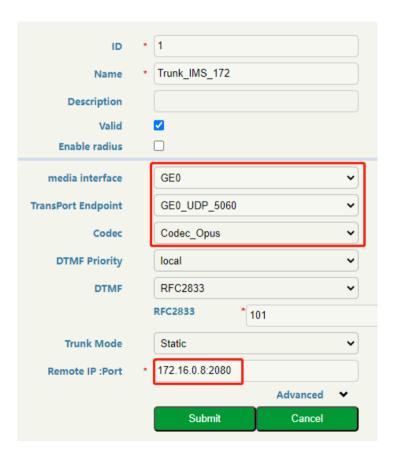
Open Page System>>>Code Profile, add two new rules.

One choose OPUS, another choose AMR.

Different IMS using different codec.



Configuration the Access SIP Trunk



Open Page System>>>Access SIP Trunk, add a new turnk.

This one for make are trunk with IMS_172, the remote IP: port input the target IMS information 172.16.0.8:2080
And this trunk will use OPUS only.



Configuration the Access SIP Trunk

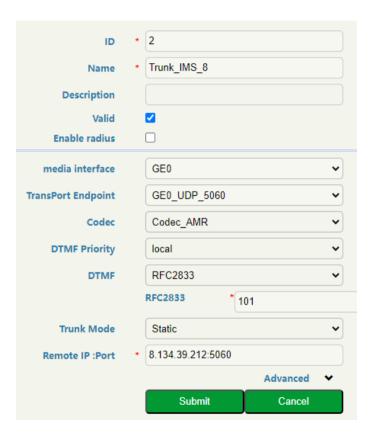
Registration	☑	
	Username * 7551000	
	Authentication ID* 7551000	
	Password *	•
	Registered Interval	
	Timeout coefficient	

The Trunk to IMS_172 should registered with sip account 7551000 Click the advanced, Enable the option "Registration", input the SIP account information.

Note: the Registered Interval should changed to 1800, mostly IMS had limited of this.



Configuration the Access SIP Trunk



Open Page System>>>Access SIP Trunk, add a new turnk.

This one for make are trunk with IMS_8, the remote IP: port input the target IMS information 8.134.39.212:5060
And this trunk will use AMR only.

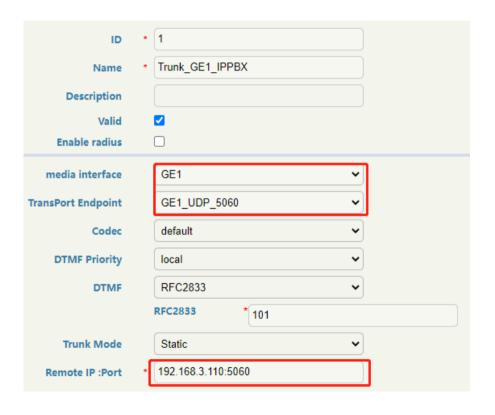


Configuration the Access SIP Trunk

Access SIP Trunk							
ID	Name	Description	TransPort Endpoint	Codec	Peer IP		
1	Trunk_IMS_172		GE0_UDP_5060	Codec_Opus	172.16.0.8:2080		
2	Trunk_IMS_8		GE0_UDP_5060	Codec_AMR	8.134.39.212:5060		



Configuration the Core SIP Trunk



Open Page System>>>Core SIP Trunk, add a new rule.

This one for make are Trunk from SBC to IPPBX.

The TransPort Endpoint choose GE1_UDP_5060

The Remote IP:port input IPPBX's 192.168.3.110:5060



Check the SIP Trunk Status

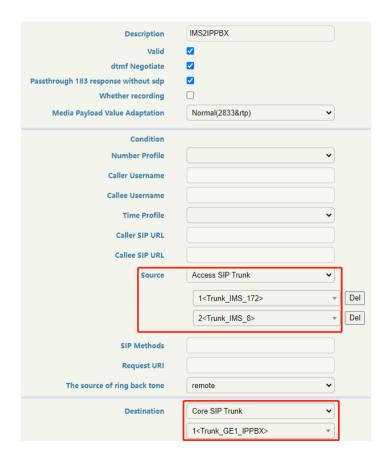


Open Page Overviewer>>>Access Trunk Status or Core Trunks Status, we can saw the trunk status is true or false.

If false, maybe the option not work or register failed or network issue.



Configuration the incoming call route



Open Page System>>>Routing Profile>>>>CallRouting, add a new rule:

The Source choose Access SIP Trunk "Trunk_IMS_172" and "Trunk_IMS_8"

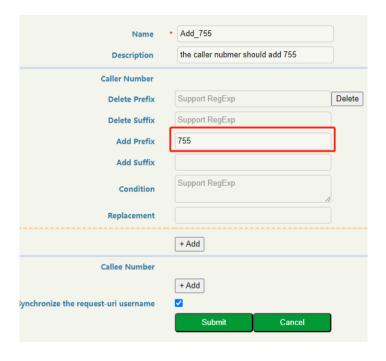
The Destination choose Core SIP Trunk "Trunk_GE1_IPPBX"

This one is incoming call route: make are route from two IMSs to IPPBX.

All call hit SBC's GE0 (172.19.10.143:5060) from two IMSs , will forward to IPPBX (192.168.3.110:5060) through interface GE1 (192.168.3.100:5060)



Configuration the outgoing call route



Open Page System>>>Number Manipulation, add a new rule:

The Add Prefix input 755.

This one mean add prefix 755 to caller number.



Configuration the outgoing call route



Open Page System>>>Routing Profile>>>>CallRouting, add a new rule:

The Source choose Core SIP Trunk "Trunk_GE1_IPPBX" The Destination choose Access SIP trunk "Trunk_IMS_172"

The Caller Username input regular chars ^10\d\d\$
The Number Manipulation choose "Add_755"

The IMS_172 have DOD 7551000-7551099, the IPPBX side have extension 10xx.

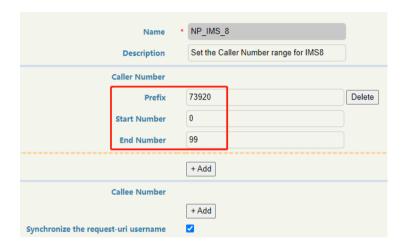
So when the extension 10xx callout, it should pass the IMS_172 and the caller should using 75510xx

Note:

The Priority should input 1000, because we need this rule high priority than another.



Configuration the Number Pool



Open Page System>>>Number pool, add a new rule: The Prefix input 73920 The Start Number set to 00 The End Number set to 99

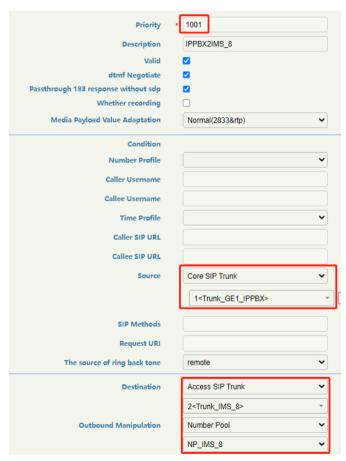
This one will set the caller number range. If the caller number in this range already, like 7392045, pass.

If the caller number out of this range, will be replaced are random one in this range. For example, the original Caller is 7559999, will be changed to 73920xx



How to Configuration

Configuration the outgoing call route



Open Page System>>>Routing Profile>>>>CallRouting, add a new rule:

The Source choose Core SIP Trunk "Trunk_GE1_IPPBX"
The Destination choose Access SIP trunk
"Trunk_IMS_8"

The Outbound Manipulation choose Number Pool and select "NP_IMS_8".

This one mean if the IPPBX send call to IMS_8, the Caller number should in range 7392000-7392099

Note:

the Priority should input 1001, because we need this rule low priority than before one.



How to Configuration

Configuration the call routing

Call Routing			
Priority	Description	Condition	Destination/Manipulation Rule
1000	IPPBX2IMS_172	^10\d\d\$, 1 <trunk_ge1_ippbx></trunk_ge1_ippbx>	1 <trunk_ims_172> / Add_755</trunk_ims_172>
1001	IPPBX2IMS_8	1 <trunk_ge1_ippbx></trunk_ge1_ippbx>	2 <trunk_ims_8> / NP_IMS_8</trunk_ims_8>
1022	IMS2IPPBX	1 <trunk_ims_172>,2<trunk_ims_8></trunk_ims_8></trunk_ims_172>	1 <trunk_ge1_ippbx> / -</trunk_ge1_ippbx>

When the call from IPPBX:

If the caller number is 10xx, will out to IMS_172 and codec using OPUS, the caller replace to 7551000-7551099. Another call to IMS_8 and codec using AMR.

If the caller number is 7392000-7392099, pass to IMS_8 directly.

If the Caller number is others, will be replaced are random number 7392000-7392099

When the call from IMS_172 and IMS_8, forward to IPPBX directly.

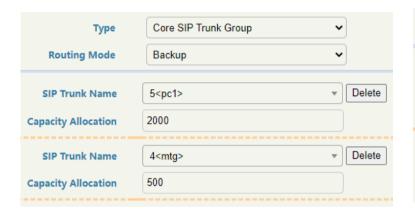
When the call form another IP, reject directly.

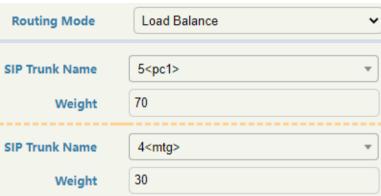






Trunk Group





Backup mode

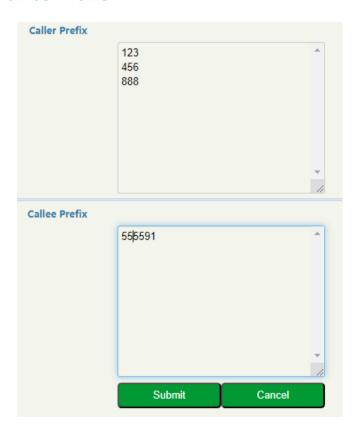
If the Capacity Allocation value same, all traffic to primary one, unless the trunk disconnect. If the Capacity Allocation value different, all traffic to primary one till it full

Load Balance mode

The SBC will base weight to dispatch call to different trunk, like Trunk A received 70%, other one 30%



Number Profile



The Number Profile for filter the calling/called number. Some times, the client should limited the number in router. You can set the prefix or whole number.

Note: This one not support Regular

This one used in Call Route level.



Black/While List

Whitelist Group	* 2235011	
Description	2235011[41-60]	
Number Description	223501160	Delete
Number Description	22350114[1-9]	Delete
Number Description	22350115[0-9]	Delete

The Black/White list use for Access Trunk and Access Network level.

Limit the external register/call to SBC, decrease the heavy of SBC and IPPBX.

Note: This one not support Regular



Number Pool

Caller Number						
Prefix	6812	Delete				
Start Number	400					
End Number	600					

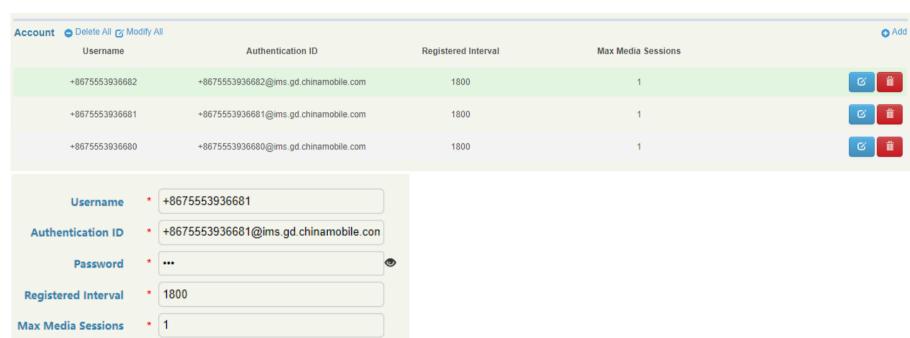
The Number Pool used for Call Route.

When SBC call out, it will replace the original number to one that in number pool.

This one used for IMS that give your one SIP account and multiple DOD.



SIP account



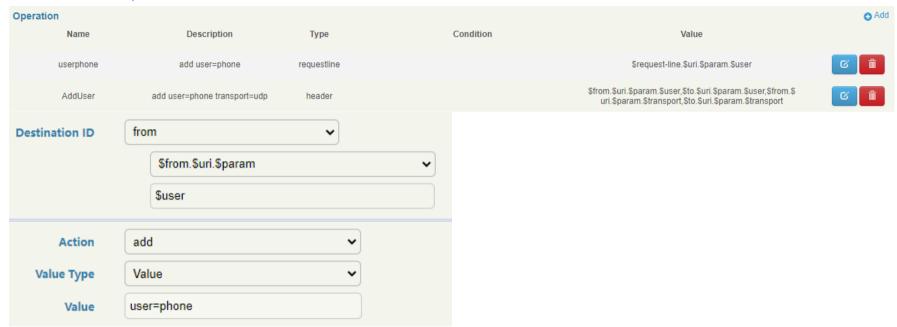
The SIP account for poll register.

In the Trunk side, we can set ONE sip account only.

Some IMS give multiple SIP accounts and request register ALL.



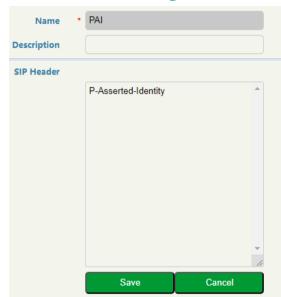
SIP Header Manipulation



SIP Header Manipulation is the real core of SBC. Target for change the SIP package.



SIP Header Passthrough



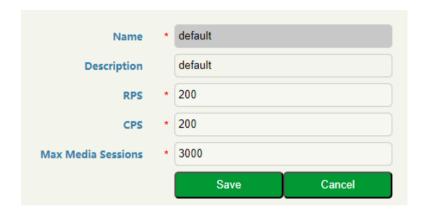
When the call passthrough the SBC, the INVITE is recombination.

SBC extraction the caller and callee number only.

But sometimes, we should passthrough some original field to destination, like P-Asserted-Identify.



Rate Limit



We can set the rate limit for SIP Trunk.

RPS: Max SIP Register request per seconds from remote side to this Trunk.

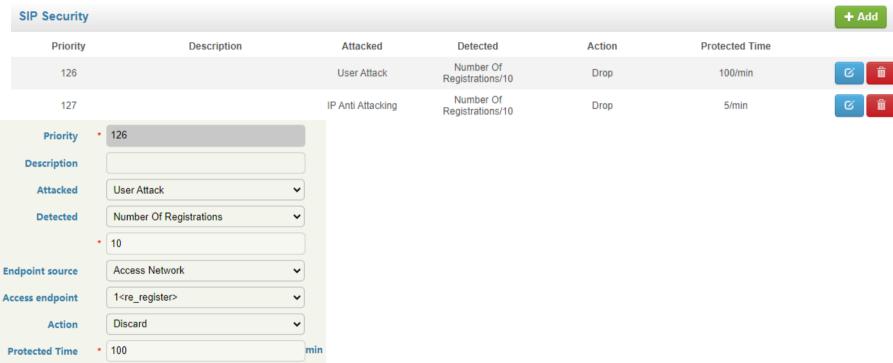
CPS: Max SIP Call request per seconds from remote side to to this Trunk.

Max Media Sessions: Max concurrent call in this trunk.

Note: the value should not overtop the license of SBC



SIP Security



SIP Security for protect the SBC and block the SIP attack.

Note: For SIP level only.







SSH of SBC

```
login as: admin
  admin@172.19.211.141's password:
BusyBox v1.22.1 (2019-01-21 11:22:54 CST) built-in shell (ash)
Enter 'help' for a list of built-in commands.
          Welcome to Command Shell
      WARNING: Authorised Access Only
  enable
admin@SBC3000 enable# show version
  Linux Version
                  : 3.4.39
  Hardware
                    0x2
  Root Image
                    38
  Boot Image
                  : 16
  App Package
                  : 2.90.4.2psp6
      BuildTime
                    2022-08-23 10:22:21 CST
      Builder
                  : root
     Md5Sum
                  : d2551be1f43ab0432f19734ee6f8a8a8
```

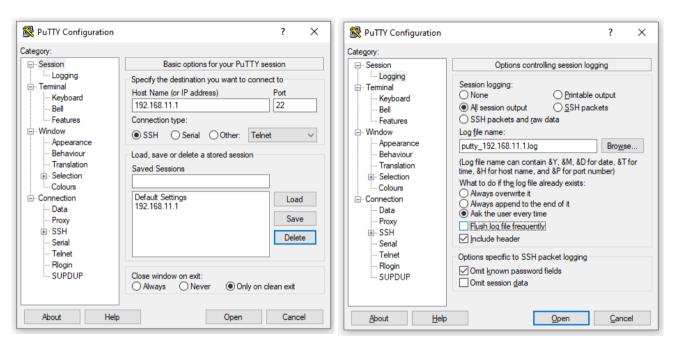
SSH is the common and safe way for troubleshooting. We can SSH the SBC through IP, the login information same of WEB login, default is admin/admin@123# After login the CLI, we can check the device IP, version, and call statistic, monitor the call, etc.

Note:

- Only Admin port enable the SSH authority as default.
- 2. SBC3000 default port is GE1



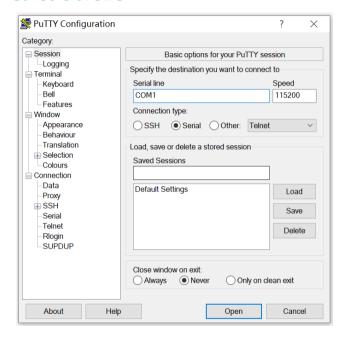
SSH of SBC



At mostly, we use application putty.exe for troubleshooting that's light and easy to use. And can save the log real time to extend file.



Console of SBC



If the device can't login the though the IP, we should check it through Console port. The console port speed is 115200 and login information same of WEB login. After login, input command enable, you will go to the same login interface of SSH. Note: If the SBC forgot password, the only is Flash whole device.



Enable Mode

```
enable
admin@SBC3000 enable# show interface
         Link encap: Ethernet HWaddr F8:C7:23:24:1E:AA
eth0
         inet addr:172.19.211.141 Bcast:172.19.255.255 Mask:255.255.0.0
         inet6 addr: fe80::fac7:23ff:fe24:leaa/64 Scope:Link
         UP BROADCAST RUNNING MULTICAST MTU:1500 Metric:1
         RX packets:73390697 errors:0 dropped:1086 overruns:0 frame:0
         TX packets:10324780 errors:0 dropped:0 overruns:0 carrier:0
         collisions:0 txqueuelen:1000
         RX bvtes:2469184967 (2.2 GiB) TX bvtes:282274978 (269.1 MiB)
         Interrupt:114
ethl
         Link encap: Ethernet HWaddr F8:C7:23:24:1E:AB
         inet addr:10.100.10.11 Bcast:10.100.255.255 Mask:255.255.0.0
         UP BROADCAST MULTICAST MTU:1500 Metric:1
         RX packets:15478328 errors:0 dropped:0 overruns:0 frame:0
         TX packets:0 errors:0 dropped:0 overruns:0 carrier:0
         collisions:0 txqueuelen:1000
         RX bytes:928699680 (885.6 MiB) TX bytes:0 (0.0 B)
         Interrupt:114
```

en go to the enable modesh int show current networksh ver show curent version detail

Note: **sh** equal **show int** equal **interface**All command support abbreviation



Enable Mode

```
admin@SBC3000 enable#
              Enter ada mode
 ada
 board
              Set board state
 configure
              Enter configuration mode
 exit
              Exit privileged mode
 fwd
              Set fwd debug
 heart
              dma ping pong
 log
              Set log level
              mem free status stack
 mem
              Negate a command or set its defaults
 no
 ping
              Send messages to network hosts
              Send messages to network hosts
 ping6
 pprof
              pprof mem cpu qc
              Give details of current processes
 ps
 quit
              Ouit from the CLI
 reboot
              Restart system or board
 reset
              Restart netswitch or other ...
 service
              Service command
 show
              Show running system information
 ssh
              SSH board or subborad
              telnet subborad
 telnet
 top
              Give details of current processes
              Trace command
 trace
              Print the route packets trace to network host
  traceroute
              Print the route packets trace to network host
  traceroute6
              upload log memprofile
 upload
```

? Quest Mark show all the support command in list.



Major command

show clock show system current time and running time

show call statisticshow the device call statisticShow call infoshow current call information

show DSP info show the device current DSP information.

show endpoint callstat show all trunk status

show register info show user register status **show device** show device type and SN

show error show the log of system error

show flashshow the occupy of Flash.show board stateshow userboards statusshow netstatshow network port detailshow serviceshow system service status

top show the process occupy status

ps show device process



Trace log

```
admin@SBC3000 enable# trace

all Enable all trace

board Turn on board tracce

call Set sip call trace parameters

level Set trace level

register Set sip register trace parameters

transport Set sip transport message trace parameters
```

Trace for enable the trace, it have different type:

all for print call trace, this one is using normal. For example, **trace all**

level for set the trace level, we have disable/emerg/alert/crit/err/warning/notice/info/debug/detail

register for set the sip register log, it have 3 parameters: User name, Access Network ID, Core Trunk ID

For example, trace register sunny 1023 *

call For set sip call log, it have 4 parameters: CallingNum, CalledNum, InboundTrunkID, OutboundTrunk ID.

For example, trace call sunny 98765 10 *

transport For set transport log, it have 6 parameters:

Transport protocol, Source IP and port, Destination IP and port, CallingNum, CalledNum, SIP model.

For example, trace transport udp 10.10.10.10:5060 20.20.20:5062 sunny 123456 INVITE

Note, if you forgot the parameter, you can use * to replace it.

trace call sunny 98765 10 * or trace transport udp 10.10.10.10:5060 * sunny 123456 *



Trace log

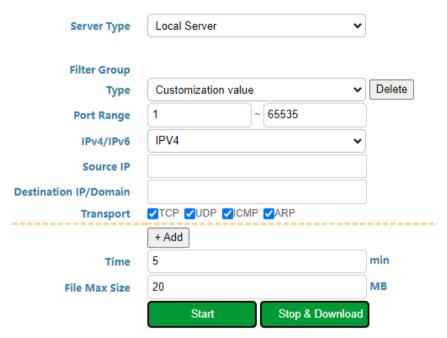
ada After input the trace command completed, should go to the ADA mode that will print the log out.

No trace all For disable the tracde

exit for exit the ada mode



Capture Call



We can do capture in device GUI.

Default click the button "start" only.

If the SBC have much concurrent call, we need use IP to filer the call.



Capture log



Capture log is very useful way also.

When the SBC have high concurrent call, we prefer to do this mode for go call log.

Note:

please choose the Level to **Debug**













support@dinstar.com



www.dinstar.com



+86 755 6191 9966