

BLOX ESBC

User Manual

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About this manual

This manual describes the Blox Esbc product application and explains how to work and use its major features. It serves as a means to describe the user interface and how to use it to accomplish common tasks.

Document Conventions

In this manual, certain words are represented in different fonts, typefaces, sizes, and weights. This highlighting is systematic; different words are represented in the same style to indicate their inclusion in a specific category. Additionally, this document has different strategies to draw User attention to certain pieces of information. In order of how critical the information is to your system, these items are marked as a note, tip, important, caution, or warning.

Icon	Purpose
	Note
	Tip/Best Practice
	Important
	Caution
	Warning

- **Bold** indicates the name of the menu items, options, dialog boxes, windows and functions.
- The color [blue](#) with underline is used to indicate cross-references and hyperlinks.
- Numbered Paragraphs - Numbered paragraphs are used to indicate tasks that need to be carried out. Text in paragraphs without numbering represents ordinary information.
- The Courier font indicates a command sequence, file type, URL, Folder/File name e.g. www.blox.org

Support Information:

Every effort has been made to ensure the accuracy of the document. If you have comments, questions, or ideas regarding the document contact online support: support@blox.org

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1. Freeblox (User Interface)

Freeblox is the GUI designed for Blox ESBC & user can configure the features and the SBC administration.

Accessing the Web GUI of Blox Esbc (follow instruction as per Quick Installation Guide)

1.1 Accessing the Web GUI

Blox Esbc Web GUI can be accessed via Management Port, Please follow the quick installation guide for details.

The message will prompted the connection is untrusted, Click on Add Exception to continue the process. Once get the certificate, to confirm security Exception and proceed to access the GUI Login page.



The WebUI has been made accessible only via HTTPS. The recommended browser for accessing Blox Esbc WebUI is Mozilla Firefox.

On launching the Blox Esbc WebUI, the web application will prompt to enter the administrator credentials to login.

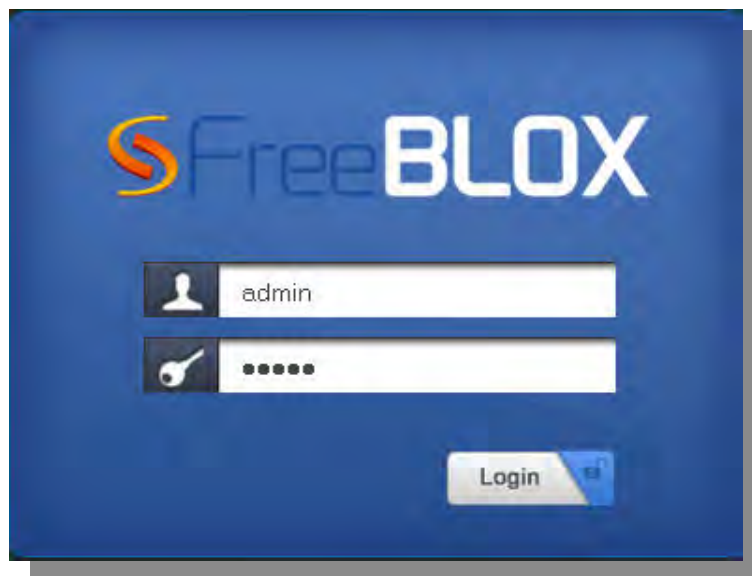


Figure 1: Login Page

End User License

In proprietary software, an end-user license or software license agreement is the contract between the licensor and purchaser, establishing the purchaser's right to use the software.

An End User License Agreement (EULA) is a legal contract between a software application author or publisher and the user of that application.

The user should be prompted to accept the Freeware license agreement and click “Agree button” to proceed further.

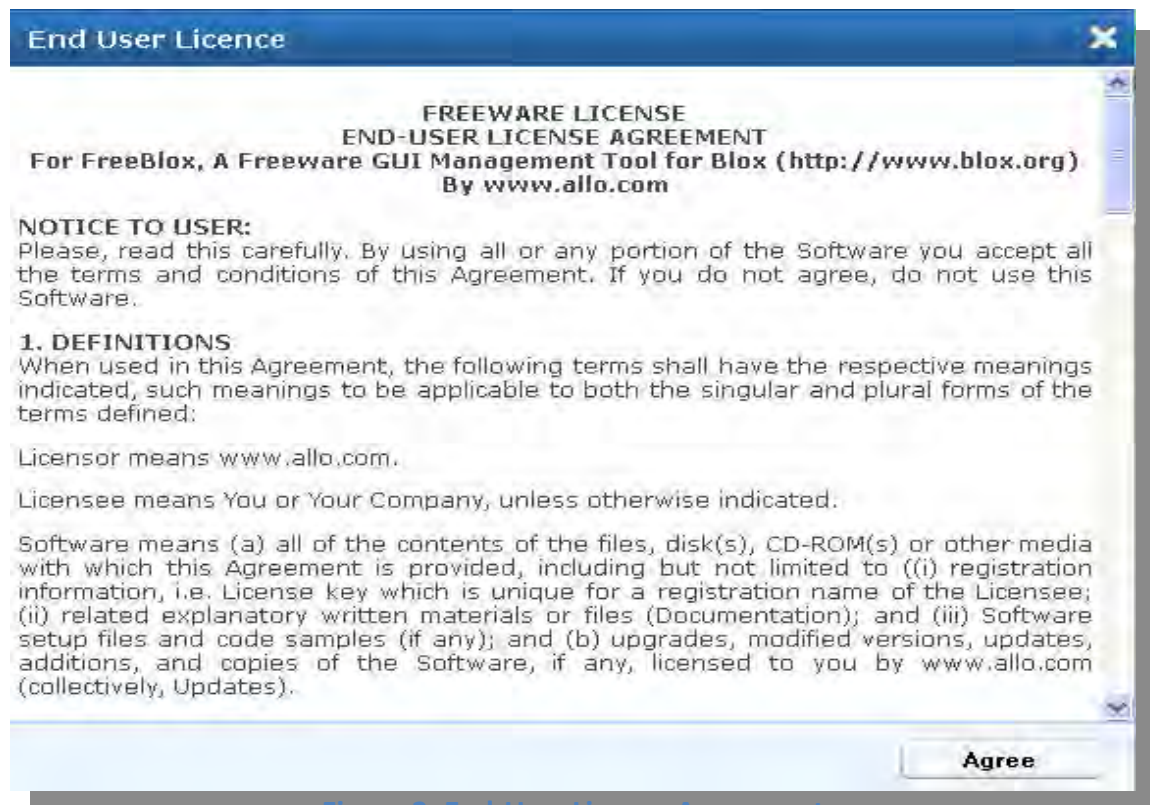


Figure 2: End-User License Agreement

1.2 Dashboard

On the very first login, the WebGUI will provide you an overview of Blox Esbc configuration status.

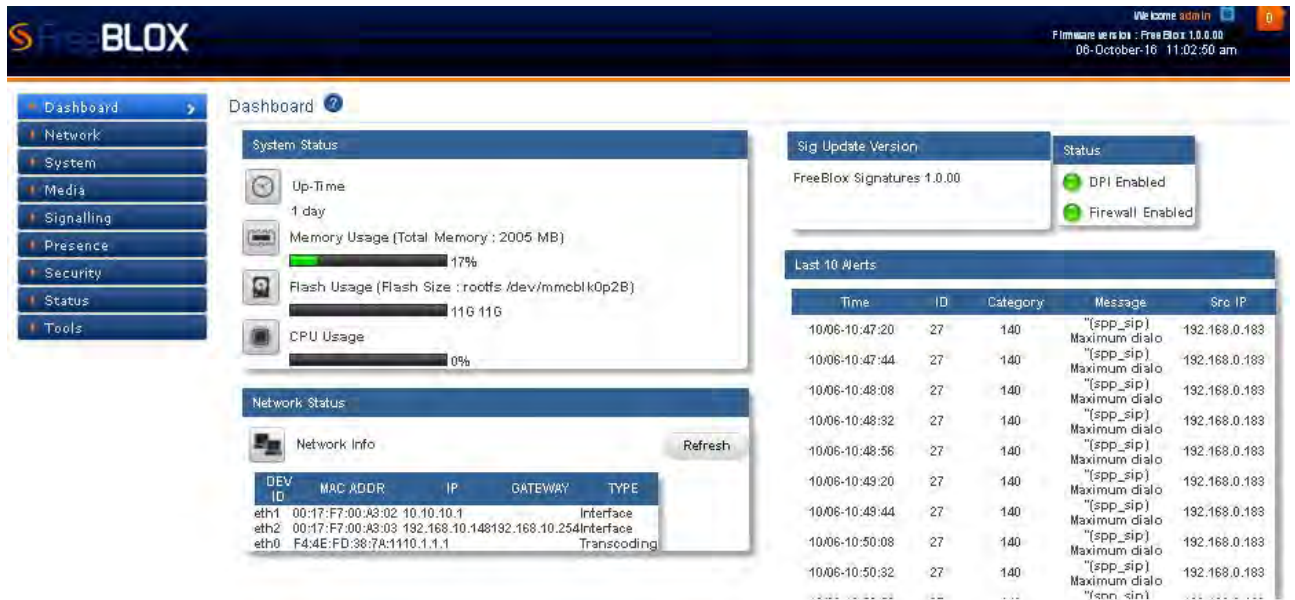



Figure 3: Dashboard

The Right of the top panel shows the current time of device. Top panel also shows the firmware release version and has an icon  which will refresh the page.

On the right side of the top panel, clicking on settings icon shows the menu which has Web Settings and Logout options.

Web Settings allows user to change the Old Password & Session Timeout values where as user name is Read-Only.

Clicking Logout will kill the session and redirects to the Login page.

System Status Panel shows Device up time, Memory Usage, Flash Usage & CPU Usage.

Sig Update Version Panel shows Blox Esbc Signature version and Release State.

Network Status Panel shows IP, LAN MAC, WAN MAC and Gateway of the device.

1.3 Network Status

Once the GUI is accessed user can configure the network configuration by following the below steps

1. After login user has to click on the refresh button in the network status in the Dashboard
2. Once the network status is refreshed it will display all the available interfaces in the board
3. Once the interfaces are displayed in the dashboard user can configure the WAN and LAN ip address of the Blox Esbc through Network--> Settings page

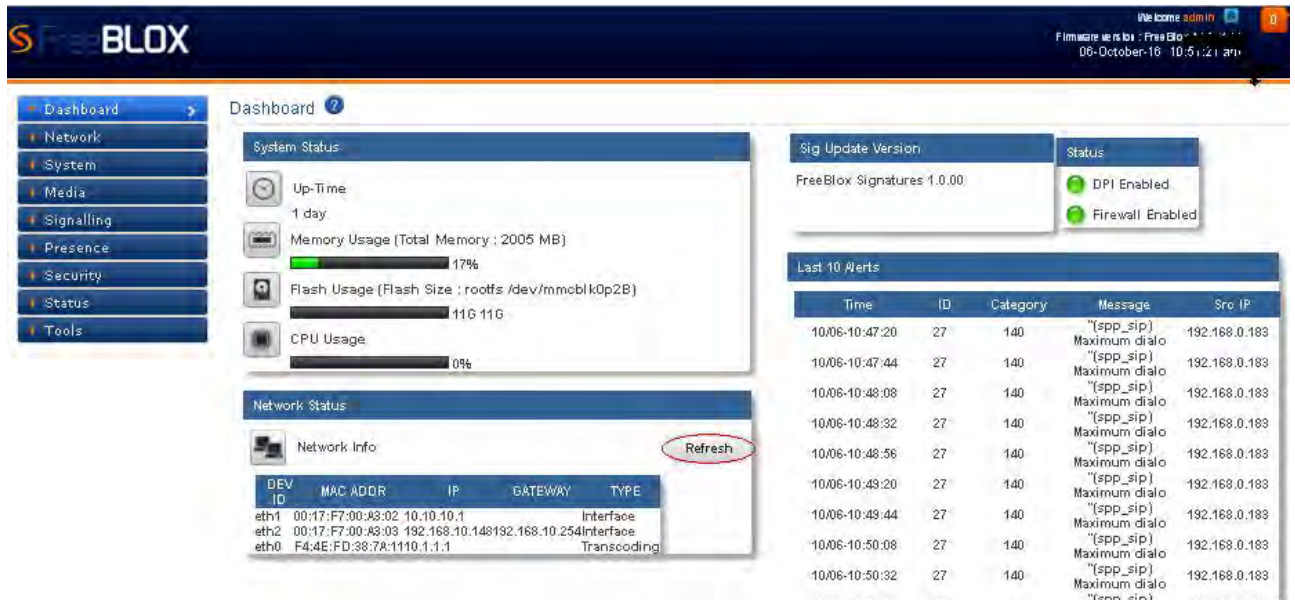


Figure 4: Network Status

Security Alert Summary Panel shows four links and on mouse over shows the details of Top 10 Signatures, Top 10 Categories, and Top Source & Top Destinations.

2. Network

This Network tab provides detailed information about Interfaces, settings, Routes and Device Access of the Blox Esbc.

In Blox Esbc mainly consists of 3 Ethernet interfaces such as internal, external and Transcoding interface.

User can configure the Virtual IP and VLAN for Blox Esbc.

2.1 Interfaces

An interface is a shared boundary across which two separate components of computer system exchange information. User can configure IP addresses for the networks.

2.1.1 Settings

Navigate through **Network > Settings**

A LAN interface deals with any type of SIP signaling which goes in and out of the Blox Esbc.

The signaling interfaces on the Blox Esbc are the physical Ethernet adapters.

It allows user to configure Host Name, IP Configuration in Static mode. IP Address/Mask, Gateway & DNS fields are editable only in Static IP mode. It also allows user to enable or disable SSH access to device.

User can save configurations by clicking on Save and can ignore saving the configurations by clicking on Cancel.

Figure 5: Settings

Transcoding Settings

If the Transcoding card is detected, user can configure the Transcoding Interface. It shows the interface type as Transcoding Interface. And also user can specify the IP address and Netmask.



Transcoding card is mandatory for SRTP and T38.

IP Troubleshooting:

In most installs, the network cards and IP settings will work straight out of the box. However, getting the network up the first time can be an exercise in frustration in some circumstances. Issues include;

- Network card compatibility
- Invalid networks settings (username, password, default gateway)
- Cable/DSL modems that cache network card hardware information

2.1.2 Virtual IP

Navigate through Network > **Interfaces**> **Virtual IP**

A Virtual IP address (VIP or VIPA) is an IP address assigned to multiple applications residing on a single server, multiple domain names, or multiple servers, rather than being assigned to a specific single server or network interface card (NIC).



Figure 6: Virtual IP

Click Add new, to create a Virtual IP.

The 'Create Virtual IP' dialog box is shown. It has a title bar with a close button (X). The form contains the following fields: 'Name' with the value 'Internal VIP'; 'Interfaces' with a dropdown menu showing 'LANIFACE' and a tooltip 'LAN / 10.10.10.1'; 'IP Address' with the value '10.2.2.1'; 'Netmask' with the value '255.255.255.0'; and 'Description' with the text 'Virtual ip address with internal interface'. At the bottom right are 'SAVE' and 'CANCEL' buttons.

Figure 7: Create Virtual IP

Create Virtual IP

Name	Specify the name for the IP address for user's reference. The user can choose any name to recognize the Virtual IP.
Interfaces	Select the appropriate interfaces from the drop down list where the user desires to create a Virtual IP. Ex: For both External/Internal can be any interface which will be configured by the user (manually). This applies to VLAN as well.
IP Address	Enter the IP address for Virtual IP settings. E.g.: 10.2.2.1
Netmask	Enter the subnet mask address for Virtual IP settings. The default setting is 255.255.255.0
Description	Provide the description for the Virtual IP. (Optional)

2.1.3 VLAN

Navigate through **Network > Interfaces > VLAN**

A VLAN is a logically separate IP sub network. It allows multiple IP networks and subnets to exist on the same-switched network. VLANs are implemented to achieve scalability, security and ease of network management and can quickly adapt to change in network requirements and relocation of workstations and server nodes.



Figure 8: VLAN

Click Add New, to create VLAN.

Figure 9: Create VLAN

Tag ID	User can specify unique Tag ID in the range of 1-4092. So that they can easily identified the multiple no of VLANs with Tag ID.
Interfaces	Select the appropriate interfaces from the drop down list where the user desires to create a VLAN. Vlan can be created for internal and external interfaces. Ex: if user wants to create the virtual IP in wan side select Eth0, WAN Interface-192.168.10.231 If the user wants to create the virtual IP in LAN side select Eth2,LAN Interface-10.0.0.1
IP Address	Enter the appropriate IP address for creating VLAN.
Netmask	Enter the subnet mask address for VLAN. The default setting is 255.255.255.0
Description	Provide the description for the VLAN. (Optional)



After clicking on 'save' button, followed by apply changes button in the top right corner of the panel.

2.2 Routes

Navigate through **Network > Interfaces> Routes**

Blox Esbc can also be used in conjunction with SIP trunks to provide call control and make routing/policy decisions on how calls are routed through the LAN/WAN.

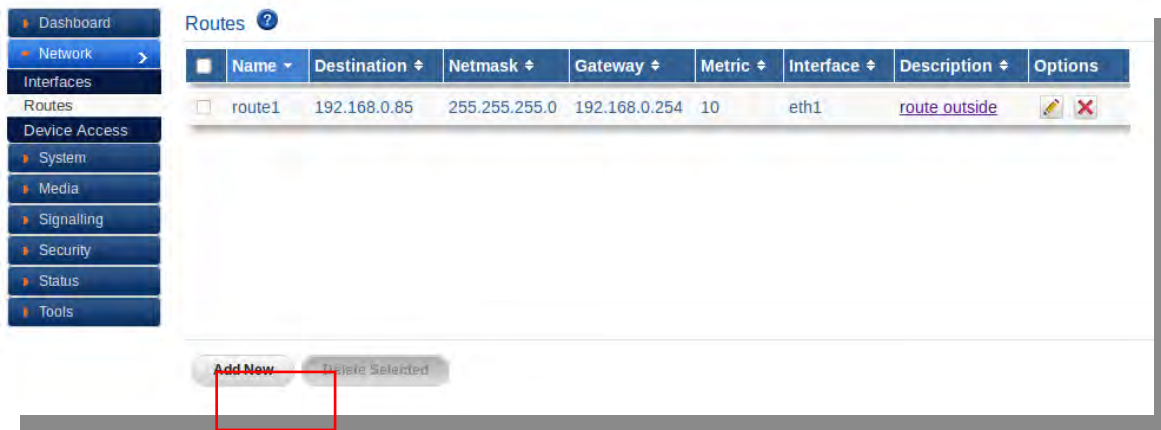


Figure 10: Routes

Click Add New, to create a route

Figure 11: Create Route

Name	Specify the name for the Routes for user's reference. The user can choose any name to recognize the Routes.
Destination	User can specify the destination Address, to where it should be routed.
Netmask	Enter the subnet mask address for Routes. The default setting is 255.255.255.0
Gateway	User can specify the gateway IP address for particular network. E.g: 192.168.0.100- IP address, the gateway will be 192.168.0.254.
Metric	User can specify Metric value in the range of 0-31

Interfaces	<p>Select the appropriate interfaces from the drop down list where the user desires to create a Route.</p> <p>Ex: if user wants to create the virtual IP in wan side select Eth0, WAN Interface-192.168.10.231</p> <p>If the user wants to create the virtual IP in LAN side select Eth2,LAN Interface-10.0.0.1</p>
Description	Provide the description for the Routes. (Optional)

2.3 Device Access

Navigate through **Network > Device Access**

It allows user to create a rule for device access that allows access to the device anywhere.

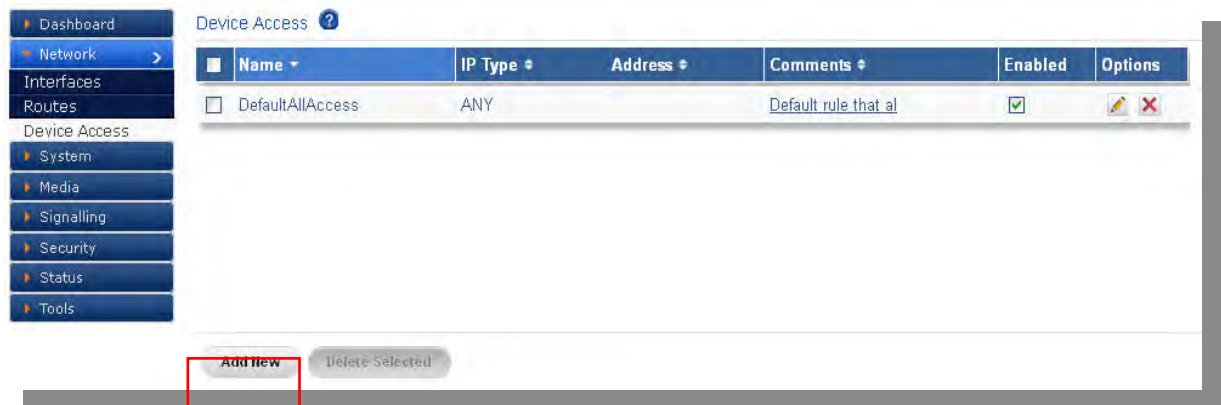


Figure 12: Device Access

Click Add New, to create Device Access Rule.

Figure 13: Create Device Access Rule

Name	Specify the name for the Device Access for user's reference. The user can choose any name to recognize the Device Access.
IP Type	User can select the appropriate IP type from the drop down list. IP types are IP_Host, IP_Network, IP_Range, and MAC_ADDR.
Address	Specify IP Address/Netmask or IP range or MAC address.
Enable	It allows the user to either enable or disable Device access rule.
Comments	User can specify the comments in the length of 64 char's.

3. System

This System tab provides detailed information about Time Settings, Logging, Package Upgrade and Email Server Settings.

3.1 Time Settings

Navigate through **System > Time Settings**

It allows user to configure Date / Time. It allows user to configure Date / Time. They can be either set manually (uses RTC) or automatically (through NTP). Default: NTP.

User can select the configuration type from the configuration type menu that allows selecting the time zones from the drop down menu.

For NTP configuration mode, add the NTP Server to the NTP list by clicking on Add button and can also delete the NTP Servers from the list by selecting and clicking on Delete button.

Clicking on Apply will apply the configurations and Cancel will ignore the configurations made.

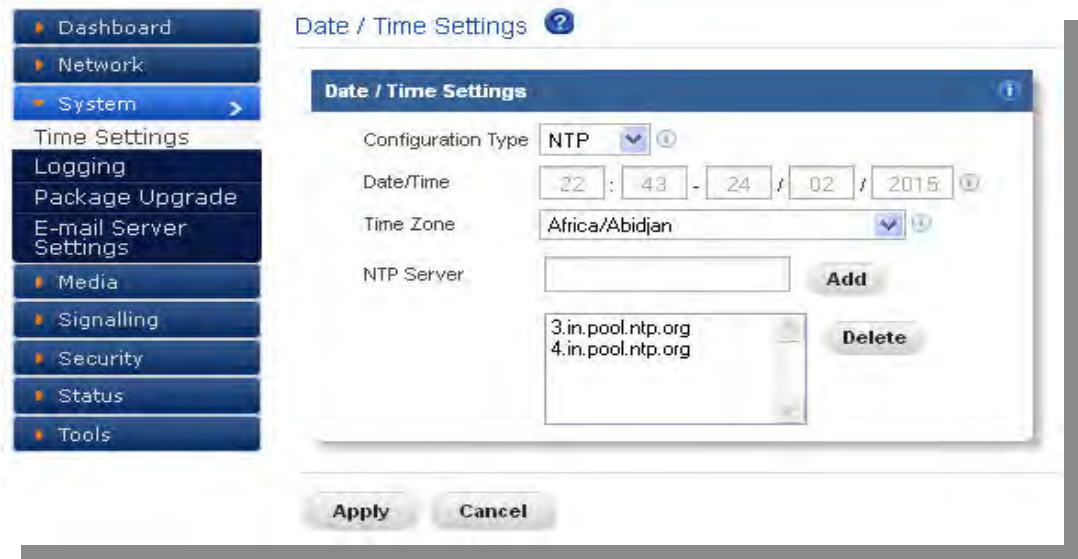


Figure 14: Date/Time Settings

3.2 Logging

Navigate through **System > Logging**

It allows user to configure Remote Log Server settings.

The administrator can configure the Blox Esbc to send the security alerts generated on detecting the SIP based attacks, to the remote Syslog server.

The logging page will allow enable/disable the remote logging of security alerts and to which Syslog server the security alerts are to be forwarded.

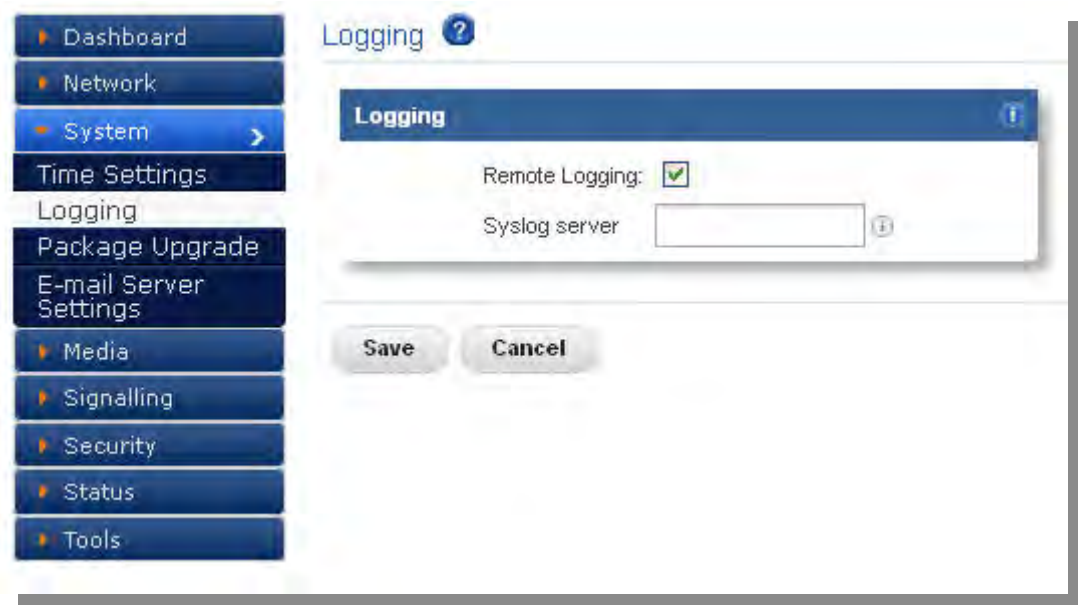


Figure 15: Logging

3.3 Package upgrade

Navigate through **System > Package Upgrade**

It can upgrade by selecting a .tgz and .iso file from the system and clicking on Upgrade button which reboots the device on success.

Click upgrade, It displays the package name, which version of upgrade is installed in the package upgrade.

Package Name	Version Installed	Platform
allo_ministun_client	0.9	x86_64
allo_mtlib	0.9	x86_64
allo_mtserver	0.9	x86_64
allo_sbc_config	0.9	x86_64
allo_sbc_miniupnpd	0.9	x86_64
allo_sbc_opensips	0.9	x86_64
allo_sbc_rtpproxy	0.9	x86_64

Figure 16: Package Upgrade

3.4 Email Server settings

Navigate through **System > Email Server Settings**

All email accounts we host, regardless of the domain name, will use the following server settings.

Figure 17: Email Server Settings

Server IP /Port	User can specify the Email server IP address and Server port.
Sender Email ID	The user can extends the verification process to include professed responsible addresses. Eg: admin@gmail.com
Receiver Email ID	The user can specify the Receiver email id Eg: roger@gmail.com
Authentication	User can select authentication from the drop down list. If authentication is required by the End point.
Username	Username of endpoint (e.g.: Testing) will use to authenticate with the Email server settings.
Password	Enter the valid password and its authenticating Email server settings.

4. Media

This section will provide detailed information about Media profile.

4.1 Media Profile

Navigate through **Media > Media Profile**

A media profile deals with all forms of media which goes in and out of the Blox Esbc. Media Profile takes care of channeling of respective media.

Through media profile, user can configure the media port range as well as type of the media like Transcoding or general media.

The media profile deal with all Transcoding functions. Example: conversion from G.729 to G.722. Also it deals with all other functions related to media (RTP/SRTP). Media profiles are the actual DSPs that perform RTP streaming, trans-coding etc.



Figure 18: Media Profile

Click Add New, to create Media Profile.

Edit Media Profile

Media Settings | Transcoding Settings

Name: TRansmedia

Description: Media profile with transcoding1

External Interface: WANIFACE (61.x.x.x) NAT

Internal Interface: LANIFACE (10.10.10.1)

transcoding Interface: TRANSIFACE (10.1.1.1)

RTP Port Start: 17000

RTP Port End: 18000

Media TOS: 11

SAVE CANCEL

Figure 19: Media Settings

Name	Enter the name for the Media Profile for user's reference. The user can choose any name to recognize the Media profile.
Description	Provide the brief description for the Media profile.(Optional)
External Interface	User can select the particular WAN IP address prompting in the drop list, which has to be sent outside of SBC.
Internal Interface	User can select the particular LAN IP address prompting in the drop list, which has to be received to internal side of SBC.
Media Interface	It specifies the Kind of media to be selected here. User can select the particular Media interface prompting in the drop list which can be either Virtual IP or a Transcoding IP.
RTP Port Start	User can be specified the starting port range which the particular media profile starts. If you want to set up the port range out of which the RTP ports will be dynamically taken, specify the End port respectively, in this field.
RTP Port End	User can be specified the ending port range which the particular media profile starts. If you want to set up the port range out of which the RTP ports will be

	dynamically taken, specify the End port respectively, in this field.
Media TOS	User can specify the Media ToS value.

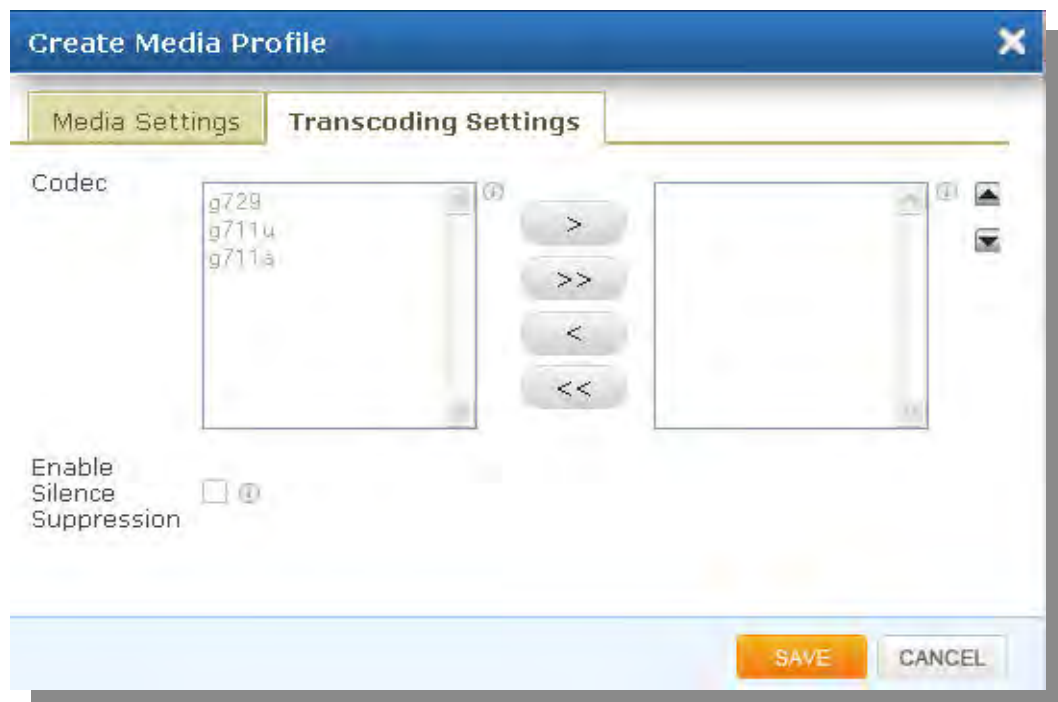


Figure 20: Transcoding Settings

Transcoding Settings

Our Transcoding cards are designed to handle complex codec translation, using dedicated DSP resources, which would otherwise be processed by host CPU in software. This card greatly reduces the MIPS or CPU consumption, so that it can be used for handling other tasks.

Codec - Our Transcoding card supports all the codecs: G722.2, AMR, GSM-EFR, GSM-FR, G.711, G.722, G.722 1C/Siren 14, G.723.1, G.726, G.729AB, T.38 FAX, iLBC

Voice signals from the PSTN come in the form of the G.711 codec, but the VoIP terminal equipment and networks can support a variety of different voice codecs, such as G.729. The VoIP infrastructure needs the capability to mediate between endpoints supporting different codecs.

User can desire to select the Codecs for Transcoding.

4.2 T38 FAX Profiles

Navigate through **Media > T38 FAX Profiles**

T38 is a protocol that describes how to send a fax over a computer data network. It is needed because fax data can not be sent over a computer data network in the same way as voice communications. T38 fax is converted to an image, sent to the other T38 fax device and then converted back to an analog fax signal.



Figure 21: T38 FAX Profiles

Click Add New, to create T38 FAX Profile.

The 'Create T38 Fax Profile' dialog box contains the following fields and options:

- Media Profile: transmedia
- Name: profile2
- Description: testing
- T38 Fax Version: 0
- T38 Max Bit Rate: 9600
- T38 Fax Rate Management: transferredTCF
- T38 Fax ECM Enable: ☒
- T38 Fax Udp EC: t38UDPRedundancy

At the bottom right are 'SAVE' and 'CANCEL' buttons.

Figure 22: Create T38 Fax Profile

Media Profile	Enter the media profile name for create T38 FAX Profile.
Name	Descriptive name for the T38 FAX Profile for user's reference.
Description	Provide the description for the T38 FAX Profile. (Optional)
T38 Fax Version	It is an ITU recommendation for allowing transmission of fax over IP networks in real time. User can select the FAX Version from the dropdown list.
T38 Max Bit Rate	It specifies the maximum bit rate from the drop down list. E.g.9600
T38 Fax Rate Management	User can select the Fax Rate Management like transferred TCP, local TCF from the drop down list.
T38 Fax ECM Enable	User can either enable or disable the FAX ECM.
T38 Fax Udp EC	User can select any types like t38UDPFEC, t38UDPRedundancy from the drop down list.

5. Signaling

Signaling section allows a user to create SIP Domain, SIP Profile, Trunk Configuration, Roaming Users, Least Cost Routing, and TLS Settings.

5.1 SIP Domain

Navigate through **Signaling > SIP Domain**

The Domain-based routing for roaming users provides support for matching an outbound dial peer based on the domain name or IP address provided in the sip domain field.



Figure 24: Create Sip Domain Profile

User can create domain names for both internal and external side and assign the domain names to corresponding sip profiles

5.2 SIP Profile

Navigate through **Signaling > SIP Profile**

A SIP Profile is an account built on the Blox Esbc which contains a set of SIP attributes that are associated to the Blox Esbc itself. The SIP profile is used as a configuration for how the external endpoints may connect to the Blox Esbc. You bind an IP address, port, and other SIP related parameters to a SIP profile.

It contains SIP UA configuration. Blox Esbc can be configured to behave as multiple UA each with a different configuration (and therefore a different set of IP: port pair each).

SIP Profile ?

	Name ▾	Interface ▾	SIP Port ▾	SIP Protocol ▾	Description ▾	Options
<input type="checkbox"/>	External profile	wan	8061	udp	External sip profile	
<input type="checkbox"/>	internal profile	lan	6060	udp	Internal sip profile	
<input type="checkbox"/>	lanprofile	lan	5060	udp	lan profile	
<input type="checkbox"/>	wanprofile	wan	8060	udp	wan profile test	

Add New Delete Selected

Figure 24: Sip Profile Results

Click Add New, to create SIP Profile.

Edit SIP Profile

Name

internal profile

Description

Internal sip profile for roaming user

Interfaces

lan LAN / 10.10.10.1 NAT

SIP Protocol/Port

udp 6060

Required TLS

Verified TLS

Server Certs

None

SIP TOS

Sip Domain

example.org example.org

Allow (IP:PORT)

Add

10.10.10.200:5060

Delete

SIP Headers

SAVE

CANCEL

Figure 25: Create Sip Profile

Name	Enter the name for the SIP Profile for user's reference. E.g.: Internal profile
Description	It provides the brief description for the profile name.(Optional)
Interfaces	<p>User can select the respective network device name from the dropdown list for internal (LAN) and external (WAN) networks.</p> <p>Ex: if user wants to create the SIP profile in wan side select network device name WANIFACE in this case or the name specified while configuring the external interface</p> <p>If the user wants to create the SIP profile in LAN side select network device name LANIFACE in this case or the name specified while configuring the internal interface</p>
SIP Protocol/Port	Blox Esbc SIP profile allows user to select multiple protocols (udp, tcp and tls) which can be available in dropdown protocol list. And Specify the SIP port in the range of 1-65535.
Required TLS	Turn on the strictest and strongest authentication possible. This parameter is used for incoming TLS connections where blox acts as a server. If disabled the verification process will succeed if the client does not provide a certificate, if enabled the verification process will only succeed if the client provides a certificate and this verifies correctly against the server's list of trusted CAs
Verified TLS	Turn on the strictest and strongest authentication possible.This parameter is used for incoming TLS connections where blox acts as a server. If disabled the blox will not request the client a client-certificate. This means that the client is not authenticated. If enabled blox sends a client-certificate request to the client.Verified TLS check box is enabled only if required TLS is checked
Server Certificates	If TLS SIP Protocol is enable, this server certificate is active. User can select the server certificates from the dropdown list.
SIP Domain	Select the appropriate domain name from the box, for the interface selected in the interfaces option
SIP TOS	The user can set the Type of Service (TOS) byte on outgoing IP packets for various protocols. The TOS byte is used by the network to provide

	some level of Quality of Service (QoS) even if the network is Congested with other traffic.
Allow (IP: PORT)	<p>Creates a list of IP addresses along with port number to be allowed for a particular SIP profile.</p> <p>E.X.: 10.10.10.200:5060</p> <p>10.10.10.300:5060</p> <p>The above mentioned IP address is internal side of Blox Esbc. User can select the respective Internal (LAN)/ External (WAN) side IP: Port or user can mention 'any' if he wants to allow all the ip address and port</p>
SIP Headers	Select the sip header manipulation rule required from the box

5.3 Sip Headers

Navigate through > **Signaling** > Sip Headers

Header manipulation is used when specific components within SIP messages need to be modified. SIP Header Manipulation provides the flexibility to add, remove, or modify any attribute in a SIP message on the Blox. The most common reason for doing this is to fix an incompatibility problem between two SIP endpoints. This could range from anything such as Softswitch/PSTN incompatibility or an issue between two different IP PBX platforms in a multi-site Enterprise where calls between them fail due to issues in the SIP messaging.

Create SIP Header Conditions

Name: Rule1

Description: header fiel removal

List of Conditions

Condition1	Action	Param1	Param2	Options
------------	--------	--------	--------	---------

Create/Edit Conditions

Condition: None

Action: remove_hf

Param-1: User-Agent

SAVE CANCEL

Figure 26: Create Header Manipulation Rule

Name	Enter the name for the Sip Header Manipulation for user's reference.
Description	It provides the brief description for the sip header manipulation name.(Optional)
Condition	Select the condition you wish to add in the SHM rule from the drop down list
Action	User can assign the action to be performed for the condition selected from the drop down list
Param	Header name to be removed

5.4 Trunk Configuration

Navigate through > **Signaling** > **Trunk Configuration**

SIP Trunks are used to connect Blox Esbc to a remote SIP Providers/User Agents. SIP Trunks can be used to communicate with SIP carriers or with IP-PBXs. It is the description of how the Blox Esbc will communicate with that endpoint. Example: IP address, port, etc.

SIP Trunks usually contains

- Remote Domain Information
- Remote authentication credentials
- Remote Registration information

SIP Trunks are bound to SIP Profiles. A single SIP Profile can be connected to multiple SIP Trunks

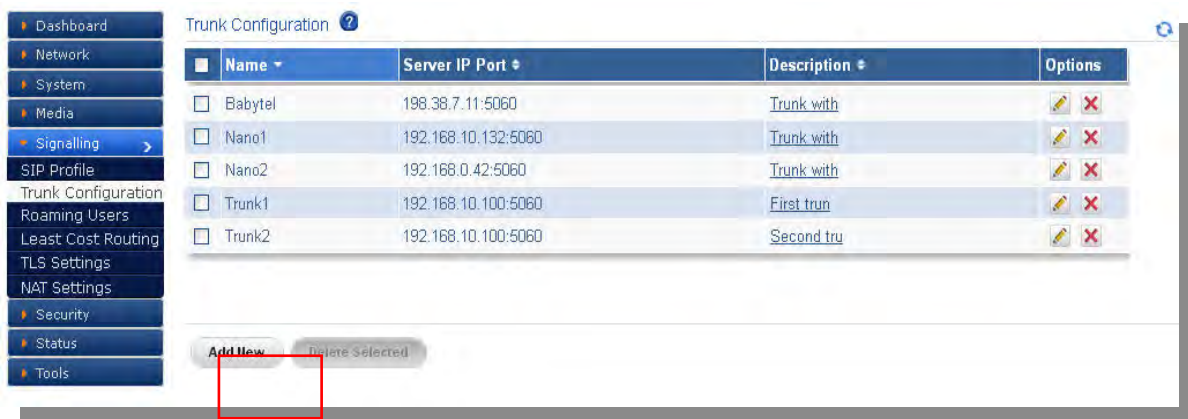


Figure 27: Trunk Configuration

Click Add New, to create Trunk Configuration.

Figure 28: Create Trunk Configuration

Trunk Name	Trunk name of the user's choice to identify for particular PBX.
Description	Provide the description for the Trunk name. (Optional)
Server, SIP (Domain/ IP: PORT)	It expects an IP address along with port number to which the particular trunk needs to be registered. E.g.: 192.168.0.200: 5060 Above example shows the IP address of provider/User agent with SIP port number.
User	The name of the user either provided by SIP provider or any extension of the PBX. E.g.: 99999 Username of endpoint (E.g.: 99999) will use to authenticate

	with the Trunk Configuration.
Password	Enter the Password and its authenticating Trunk Configuration.
SIP Registrar (IP: Port)	It expects an IP address along with port number where to get registered. Eg. 192.168.0.200:5060 It specifies the SIP registrar in the format: IP Address and port number of the PBX.
Registrar Expire	SIP Trunk registration expiry timeout, Specify Registrar expire in the range 360-3600.
Outbound Caller ID	Configure the Caller ID Number that would be applied for outbound calls over this trunk. E.g.:99999
Outbound Proxy URI	IP address or hostname with port of the outbound proxy URI. This ensures that all the SIP packets are sent via specified proxy URI. Specify outbound proxy URI in the format IP Address: Port
User Agent	Specify the customized SIP User Agent used for SIP Method, default Blox-<version>.
Internal SIP Profile	Internal (LAN) SIP Profile interfaces to the local PBX or IP end points. User can select Internal SIP profile from the dropdown list.
External SIP Profile	External SIP Profile interfaces to the ITSP or SIP trunk provider. User can select External SIP profile from the dropdown list.
Media Profile	In this field, user can select the type of media like Transcoding or general. E.g.: general media- 10.2.2.1 Transcoding media- 10.1.1.1
Media Encryption (LAN)	The media encryption feature using secure RTP (SRTP) delivers the ability to encrypt LAN Side media packets. SRTP is a security profile for RTP that adds confidentiality, message authentication, and replay protection to that protocol.
Media Encryption (WAN)	The media encryption feature using secure RTP (SRTP) delivers the ability to encrypt WAN Side media packets. SRTP is a security profile for RTP that adds confidentiality, message

	authentication, and replay protection to that protocol.
T38 Profile	Provide a T38 Profile which is already configured. The drop down menu will show the available T38 profiles.
Add Prefix	It's an optional field, in which user can add a number as a prefix for the particular trunk. Specify add prefix before the dialed number.
Strip Digits	It allows user to specify the number of digits that will be stripped from the dialed number. E.g.: 5- It will get stripped from the caller no.
Allow Inbound	It provides a checkbox that user can enable/disable the option allow inbound for trunk configuration.
Inbound URI	Provide an IP address with port number of respective internal (LAN) PBX. Ex. 10.10.10.200:5060 Internal (LAN) PBX IP along with SIP port.
Max Inbound	The user can restrict the number of incoming calls, which can be coming through that particular trunk. Also user can select Max inbound to configure the trunks.
Allow Outbound	This field allows the user to either enable or disable outbound calls. User can make calls through that particular trunk.
Max Outbound	The user can restrict the number of outgoing calls, which can be making through that particular trunk.








5.5 Roaming Users

Navigate through > **Signaling** > **Roaming Users**

Roaming user is to create a profile for Internal (LAN) PBX such that user agents can register from the External network providing the details of roaming profiles.

Roaming user is a kind of user/extension which can register to the LAN side PBX by giving the Blox Esbc IP address and roaming port during registration

Roaming Users


	Name ▾	Force Expire ▴	Description ▴	Options
	Roam2	3600	Roaming user with Na	 
	Roamelastix	3600	Roaming user configu	 

Add New


Delete Selected

Figure28: Roaming Users Result


Click Add New, to create Roaming user Profile.

Create Roaming User Profile 


Name

Roaming User 


Description

Roaming user for PBX200 

Internal SIP Profile

internal profile ▾ 


External SIP Profile

External profile ▾ 


Sip Domains Destination URI

example.org 10.10.10.200:5060


Media Profile

GeneralMedia ▾ 


Media Encryption (LAN)

None ▾ 



Media Encryption (WAN)

None ▾ 


T38 Profile

None ▾ 


IP Auth


Force Expire

3600 



Max Inbound

100 


Max Outbound

100 


Presence


Presence Server



Presence Domain



Enum



Enum Type

enum ▾

Enum Suffix

e164.arpa

Enum Service

sip

SAVE

CANCEL

Figure 29: Create Roaming User Profile

Create Roaming User Profile

Name	Enter a name for the Roaming users for user's reference. The user can choose any name to recognize the Roaming User profile.
Description	Provide the description for Roaming profile.(Optional)
LAN SIP Profile	Expects internal (LAN) side IP which is placed behind the Blox Esbc. User can select the configured internal side SIP profile from the drop down list.
WAN SIP Profile	Expects External (WAN) side IP which is present in the external network. User can select the configured external side SIP profile from the drop down list.
SIP Domains/Destination URI	The domain name which user is selected in the external sip profile configuration will be displayed in the sip domains field The user has to can provide the URI of the destination for which calls has to be routed if reaches the corresponding domain name
Media Profile	In this filed user can select the type of media like Transcoding or general User can select the Media profile from the drop down list.
Media Encryption (LAN)	The media encryption feature using secure RTP (SRTP) delivers the ability to encrypt LAN Side media packets. SRTP is a security profile for RTP that adds confidentiality, message authentication, and replay protection to that protocol. This field is enabled only if the user is using transcoding media in the media profile
Media Encryption (WAN)	The media encryption feature using secure RTP (SRTP) delivers the ability to encrypt WAN Side media packets. SRTP is a security profile for RTP that adds confidentiality, message authentication, and replay protection to that protocol. This field is enabled only if the user is using transcoding media in the media profile

T38 Profile	Provide a T38 Profile which is already configured. The drop down menu will show the available T38 profiles.
IP Auth	User can check this option if need to enable ip authentication for roaming users, and configure the allowable ip:port in the allow (ip:port) field in the external sip profile configuration
Force Expire	Force a time period where the roaming user registration will be forced to expire. Specify force expire in the range of 1-3600.
Max Inbound	User can specify the max allowable roaming users in this field for inbound
Max Outbound	User can specify the max allowable roaming users in this field for outbound
Presence	User can check this option if he want to use the presence feature, blox will work as a presence client
Presence server	User can provide the presence server ip address in this field in the format IP address: Port.
Presence Domain	User can provide the presence server domain address in this field
Enum	User can check this option if he want to use the enum feature
Enum Type	Blox Esbc enum type allows user to select multiple enum(enum, isn and isn2) which can be available in dropdown list.
Enum suffix	User can specify the suffix for the enum in this field or can use the default value, default suffix will be e164.arpa
Enum service	User can specify the service which he wanted to use with enum, default is sip

5.6 Least Cost Routing

Navigate through **Signaling -> Least Cost Routing**

Least cost routing uses the graphic user interface of the Blox Esbc to allow users to create routing rules. Least Cost Routing rules can be used to route calls based on route costs.




Figure 30: Least Cost Routing

Click Add New, to create LCR Rule.



Figure 31: Create LCR Rule

Name	Enter the name of the Least Cost Routing for user's reference.
Description	Provide the description for the Least Call Routing.
Prefix to match	User can add any number as a prefix for the respective LCR rule. They can identify prefix to match in the range of (1-16) numbers only.
LAN SIP Profile	Provide the Internal (LAN) side SIP profile which has been created for LCR. Select respective SIP profile where the call need to get routed
Trunks Config	The user has to create an LCR similar for trunk configuration. By default it displays all the trunks configured in the Blox Esbc. User can select a particular trunk accordingly to their priority.  <i>The prefix has to be matched with configured prefix inside the PBX for that particular selected trunk.</i>

5.7 TLS Settings

5.7.1 Device Root CA

In this Section, user can upload a CA file and generate the same along with the Country name, Province Name, Organization name, Email address, Encryption strength and valid days etc.

To download Device root CA, user need to generate the certificate locally by using generate option.



Figure 32: Device Root CA

5.7.2 Server Certificates

In this section, user can upload the certificates with Passphrases.

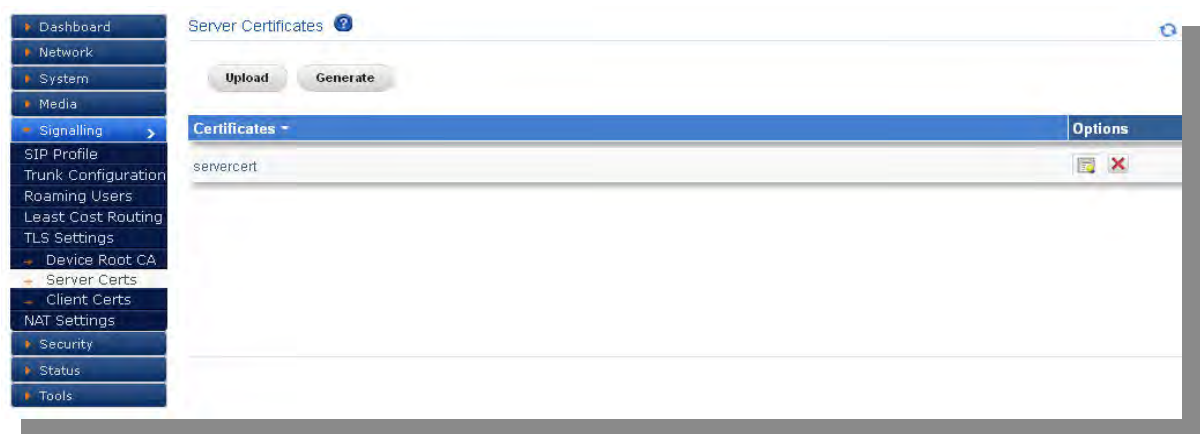


Figure 33: Server Certificates

5.7.3 Client Certificates

It provides detailed information about the client certificates with viable options which are uploaded in Blox Esbc.

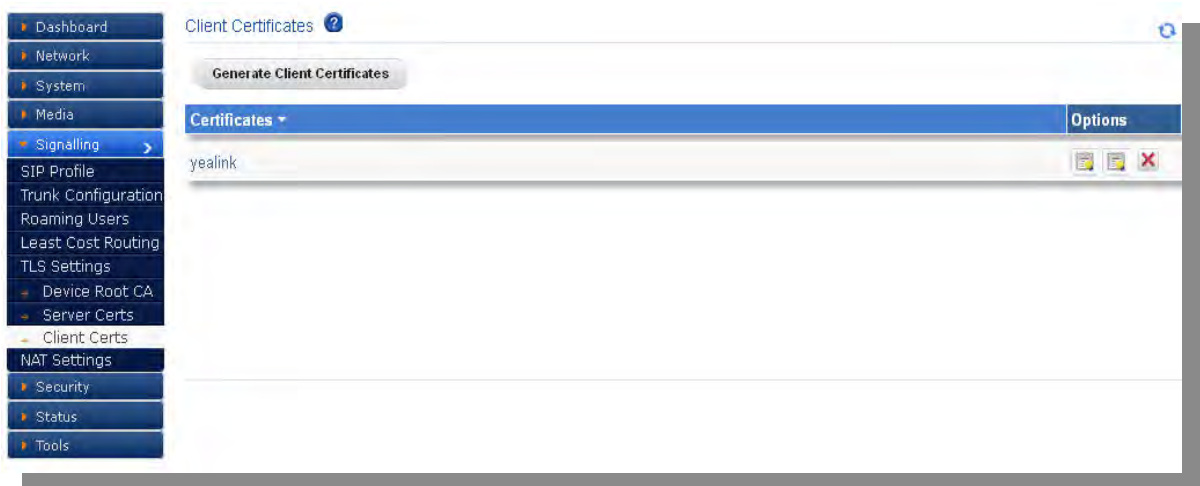


Figure 34: Client Certificates

Initially Device Root CA needs to be generated

After generating the Device Root CA, Server Certificates will get an option to generate the Server Certificates.

Similarly client certificates needs to be generated and saved locally. Client Certificate needs to be uploaded in the End user agent client (say IP Phones: yealink, snom)

5.8 General Settings

Global Settings

In global settings user can specify the User-Agent name which he wants to sent in the message going out from the Blox Esbc

User can also specify the maximum number of cdr records to be store in the Blox Esbc

Nat Settings

NAT (Network Address Translation) translates the source IP address of a device on one network interface, usually the Internal, to a different IP address as it leaves another interface, usually the interface connected to the ISP and the Internet. This enables a single public address to represent a significantly larger number of private addresses.

The screenshot displays the 'General Settings' configuration page. On the left, a vertical navigation menu lists various system components, with 'Signalling' currently selected. The main panel is divided into two sections. The 'Global Settings' section includes input fields for 'User Agent' (containing 'Blox-Esbc') and 'Maximum CDR Record' (containing '10000'). The 'NAT Settings' section includes a dropdown for 'Keepalive SIP Method' (set to 'OPTIONS'), a text field for 'Keepalive Interval' (set to '30'), and a text field for 'Keepalive From-URI' (set to 'sip: sbc@blox.org'). 'Save' and 'Cancel' buttons are located at the bottom of the settings area.

Figure 35: General Settings

6.Presence

Presence also known as presence information, conveys the ability and willingness of a user to communicate across a set of devices. SIP is particularly well suited as a presence protocol. SIP location services already contain presence information, in the form of registrations. Furthermore, SIP networks are capable of routing requests from any user on the network to the server that holds the registration state for a user. The presence support is used fo Roaming user, this support will add SIP method SUBSCRIBE and NOTIFY

6.1 Subscribers

Navigate through Presence > subscribers

Configure the subscribers used for the presence feature



The screenshot shows a window titled "Create Subscribers Profile". Inside the window, there are three labeled fields: "User name" followed by a text input box, "Operator" followed by a checkbox, and "Roaming User Profile" followed by a dropdown menu currently displaying "--No Profile--". At the bottom right of the window, there are two buttons: "SAVE" (highlighted in orange) and "CANCEL".

Figure 36: Subscribers

User name	Enter the user name of the subscriber you wish to use
operator	Check the box if the user want to use the subscriber as operator
Roaming user Profile	Select rhe required roaming user profile which the subscriber want to use from the drop down box

6.2 Events

Navigate through Presence > Events

Configure the events list for the subscriber

Figure 37: Events

From Username	Enter the username for which user need to create the event
Roaming User Profile	Select the required roaming user profile which user want to use for this event from the drop down list
Subscribers List	Select the required subscribers from the available list to active list for this event
Events	Configure this field as “ message-summary”
AC Packets	Use the default value for this field is application/pidf+xml
Expire	Configure the expire time for the event

7. Security

7.1 SIP

7.1.1 Attacks Detection

Navigate through **Security > SIP > Attacks Detection**

The SIP Attack Detection page allows to configure the SIP Deep packet Inspection rules categories. The administrator can enable/disable the inspection against particular category of rules, action to be taken on detecting attacks matching the rules in the categories.

The possible actions that the Blox Esbc can execute are log the alert, block the packets containing the attack vector and blacklist the attacker IP for the given duration. The blocking duration of how long the attacker up needs to be blocked is also configure per category level.

The following table lists the SIP Deep packet Inspection rules categories supported in Blox Esbc and configuration parameters in each category.

Category	Action	Blocking Duration (seconds)	Enabled	Options
Sip Devices Scanning	Block	120	<input checked="" type="checkbox"/>	
SIP Extensions Discovery	Block	120	<input checked="" type="checkbox"/>	
Multiple Authentication Failures/Bruteforce password cracking Attempt	Log	1800	<input checked="" type="checkbox"/>	
Ghost calls Attempt	Block	1800	<input checked="" type="checkbox"/>	
SIP Protocol Compliance	Log	none	<input checked="" type="checkbox"/>	
Sip Dos Attacks	Block	1800	<input checked="" type="checkbox"/>	
Sip DDoS Attacks	Block	1800	<input checked="" type="checkbox"/>	
TCP Syn Flood	Block	1800	<input checked="" type="checkbox"/>	
TCP Flood	Block	1800	<input checked="" type="checkbox"/>	
TCP Distributed Flood	Block	1800	<input checked="" type="checkbox"/>	

Figure 36: SIP Attacks Detection

Category	Description	User Configurable options
Reconnaissance Attacks	<p>This can be considered as the first step of attacking any system or a network. In this a hacker tries to learn information about our network typically conducts a ping sweep of the target network to determine which IP addresses are alive. Then the intruder determines which services or ports are active on the live IP addresses. From this information, the intruder queries the ports to determine the type and version of the application and operating system running on the target host.</p> <p>The attacker often uses port scanning, for example, to discover any vulnerable ports. After a port scan, an attacker usually exploits known vulnerabilities of services associated with open ports that were detected.</p>	-

SIP Devices Scanning	The intruder will scan the PBX ports to see what devices are connected to it. With that info, he can exploit 3rd party vulnerabilities. The Blox Esbc will not respond to his query.	-
SIP Extensions Discovery	The intruder will ask the PBX to divulge the range of the extension numbers. With that info, he can try different passwords to take control of these extensions. The Blox Esbc will not respond to that query.	Invalid SIP User Registration Attempts/Duration
Multiple Authentication Failures/Brute force password Attempt	The intruder will try to log in with different user names and passwords multiple times. Once he succeeds, he will have control of that extension. The Blox Esbc can block, log or blacklist the IP for a period of time if it exceeds the authorized number of trials/second.	Failed Authentication Attempts/Duration
Ghost calls Attempt	The intruder will generate calls to an extension and it will look like the calls come from that same extension. His goal is to crash the PBX resulting in disrupted communication. The Blox Esbc can block, log or blacklist the IP for a period of time if it exceeds the authorized number of trials/second.	No of Anonymous Invite Responses/Duration
SIP Protocol Compliance	This kind of attacks refers to use of some kind of automated tool like SIPP to generate false script where some of the most important fields of SIP headers and body can be modified in terms of their length like “From header length”, “To Header length”, “Contact length” . It can also be useful in handling the correct use of Maximum Dialog within a session, SIP Ports and its Protocol.	-
SIP Anomaly Attacks	The SIP Deep packet inspection engine running the	-

	<p>STM appliance has been made to inspect the SIP traffic with the SIP Security Compliance rules in built into the SIP DPI engine.</p> <p>The anomalies in the SIP Message headers can result to various erroneous conditions, SIP parser failures & malformed packets which will lead to SIP applications vulnerable to attacks.</p> <p>The Default parameters will be used by the SIP deep packet engine for identifying the different protocol anomaly conditions and take the action configured by the administrator.</p> <p>Configuring inappropriate values for these parameters can result to the disruptive impact in the VOIP deployment. Administrators with more in-depth understanding with the SIP Protocol can choose to tune these parameters for their specific deployment needs. Otherwise, it is recommended to use the default settings for these parameters.</p>	
SIP Dos Attacks	Flooding attempts using various SIP messages.	No of SIP Request Messages/Duration
SIP DDos Attacks	Distributed flooding attempts using various SIP messages.	No of SIP Response Messages/Duration
SIP Cross site scripting Attacks	<p>Cross Site Scripting (also known as XSS or CSS) is one of the most common application layer hacking techniques.</p> <p>In general, cross-site scripting refers to that hacking technique that leverages vulnerabilities in the code of a web application allow an attacker to send malicious content from an end-user and collect some type of data from the victim.</p>	-

	The use of XSS might compromise private information, manipulate or steal cookies, create requests that can be mistaken for those of a valid user, or execute malicious code on the end-user systems. It can be used to steal data about “From Header”, “To Header”, and “Call -ID”, “CONTACT “,” Extension Password and other such confidential data.	
Buffer overflow Attacks	This refers to illegally trying to access the resources of the SIP device like its memory address for which it does not have the authenticate permissions leading to data corruption of this address along with its adjacent address.	-
3 rd Party Vendor Vulnerabilities	This attack refers to any malicious activities from 3 rd party like DIGIUM Asterisk channel driver DOS attempt and other such attack.	-
TCP Syn Flood	It’s a kind of DOS attack in which a large number of TCP SYN packets are sent to the victim’s device .Each of these packets will try to establish a new session, thus consuming the victim's device resources. Such attack is also called open half connection as these new sessions are not terminated and finally the legitimate users are barred from availing the Device resources.	No of TCP Syn Packet within specified duration
TCP Flood	This refers to flooding the device with general TCP packet on any port where legitimate users are barred from availing the Device resources after some interval of time.	No of TCP Packet within specified duration
TCP Distributed Flood	In a TCP DDos attack, the incoming TCP traffic flooding the victim originates from many different sources – potentially hundreds of thousands or more. This effectively makes it impossible to stop the attack	No of TCP Packet within specified duration

	<p>simply by blocking a single IP address; plus, it is very difficult to distinguish legitimate user traffic from attack traffic when spread across so many points of origin.</p>	
UDP Flood	<p>This refers to flooding the device with general UDP packet on any port where legitimate users are barred from availing the Device resources after some interval of time.</p>	No of UDP Packet within specified duration
UDP Distributed Flood	<p>In a UDP DDos attack, the incoming UDP traffic flooding the victim originates from many different sources – potentially hundreds of thousands or more. This effectively makes it impossible to stop the attack simply by blocking a single IP address; plus, it is very difficult to distinguish legitimate user traffic from attack traffic when spread across so many points of origin.</p>	No of UDP Packet within specified duration
Generic Attacks	<p>Some of the common attacks under this category are Bye Teardown, Registration Hijack, Registration Adder, and Registration Eraser.</p> <p>1) Bye Teardown attack disrupts a call that is in session between two users.</p> <p>2) Registration Hijack: The first step in hijacking a registration is to find register able addresses and it hijacks the already registered extension.</p> <p>3) Registration Adder: This tool attempts to bind another SIP address to the target, effectively making a phone call ring in two places (the legitimate user's desk phone and the attacker's phone).</p> <p>4) Registration Eraser: This tool will effectively cause a denial of service by sending a spoofed SIP REGISTER</p>	-

	message to convince the proxy that a phone/user is unavailable.	
--	-----------------------------------------------------------------	--

7.1.2 Protocol Compliance

Navigate through **Security > SIP > Protocol Compliance**

The SIP Deep packet inspection engine running the Blox Esbc appliance has been made to inspect the SIP traffic with the SIP Security Compliance rules in built into the SIP DPI engine. The anomalies in the SIP Message headers can result to various erroneous conditions, SIP parser failures & malformed packets which will lead to SIP applications vulnerable to attacks.

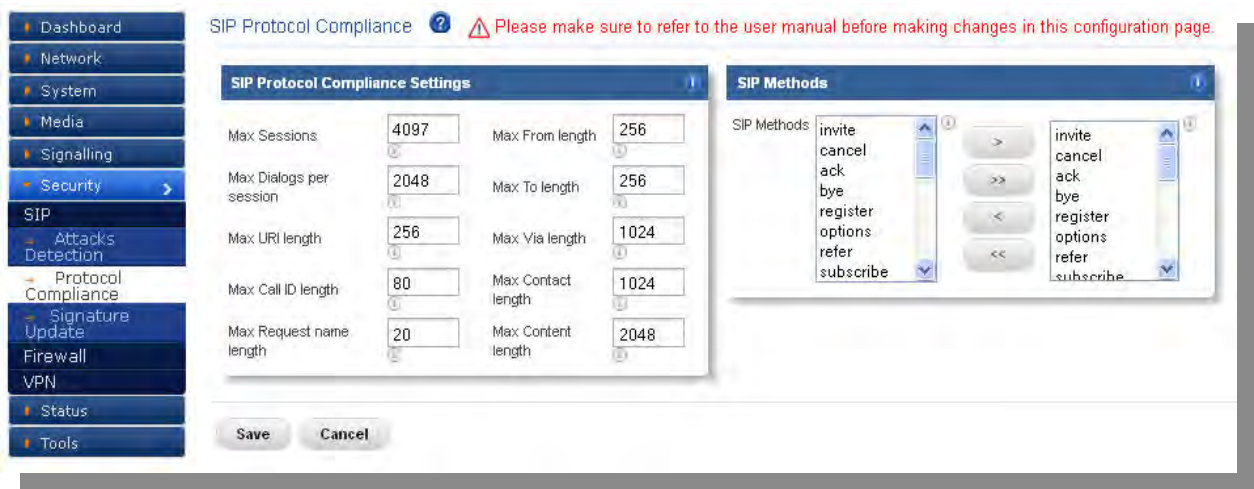


Figure 37: Protocol Compliance

The following parameters will be used by the SIP deep packet engine for identifying the different protocol anomaly conditions and take the action configured by the administrator.

Configuring inappropriate values for these parameters can result to the disruptive impact in the VOIP deployment. Administrators with more in-depth understanding with the SIP Protocol can choose to tune these parameters for their specific deployment needs. Otherwise it is recommended to use the default settings for these parameters.

Max_sessions

A SIP session is the application level connection setup created between the SIP server and SIP client for exchanging the audio/video messages with each other.

The max_sessions parameter defines the maximum number session that SIP deep packet inspection engine can keep track of. The default value has been set as 4096.

Max Dialogs per session

Max_Dialogs_per_session specifies the maximum number of SIP messages transaction that can happen between the SIP server and client.

Methods

This parameter specifies on what methods to check for SIP messages.

Following are the SIP messages that SIP DPI Engine can identify: (1) invite, (2) cancel, (3) ack, (4) bye, (5) register, (6) options, (7) refer, (8) subscribe, (9) update (10) join (11) info (12) message (13) notify (14) prack.

Max_uri_len

The URI identifies the user or service to which SIP request is being addressed. Max_uri_len specifies the maximum Request URI field size. Default is set to 256. The allowed range for this option is 1 - 65535.

Max_call_id_len

The Call-ID header field in SIP message acts as a unique identifier that relates to sequence of messages exchanged between SIP client and server. Max_call_id_len specifies the maximum Call-ID field size. Default is set to 256. The allowed range for this option is 1 - 65535.

Max_requestName_len

Max_requestName_len specifies the maximum request name size that is part of the CSeq ID. Default is set to 20. The allowed range for this option is 1 - 65535

Max_from_len

The From header field indicates the identity of the initiator of the SIP request. Max_from_len specifies the maximum From field size. The allowed range for this option is 1 - 65535.

Max_to_len

The To header field specifies the desired recipient of the SIP request. Max_to_len specifies the maximum To field size. Default is set to 25

6. The allowed range for this option is 1 - 65535.

Max_via_len

The Via header field indicates the transport used for the SIP transaction & identifies the location where the SIP response is to be sent.

Max_via_len specifies the maximum via field size. Default is set to 1024. The allowed range for this option is 1 - 65535.

Max_contact_len

Identifier used to contact that specific instance of the SIP client/server for subsequent requests.

Max_contact_len specifies the maximum Contact field size. Default is set to 256. The allowed range for this option is 1 - 65535.

Max_content_len

Max_content_len specifies the maximum content length of the message body. Default is set to 1024. The allowed range for this option is 1 - 65535.

6.1.3 Signature Update

Navigate through **Security > SIP > Signature Update**

It allows user to schedule the update by configuring the time schedule fields. Apply will cause signature to be updated according to the time scheduled by user.

The option “Update Signatures now” updates the signatures at that moment.

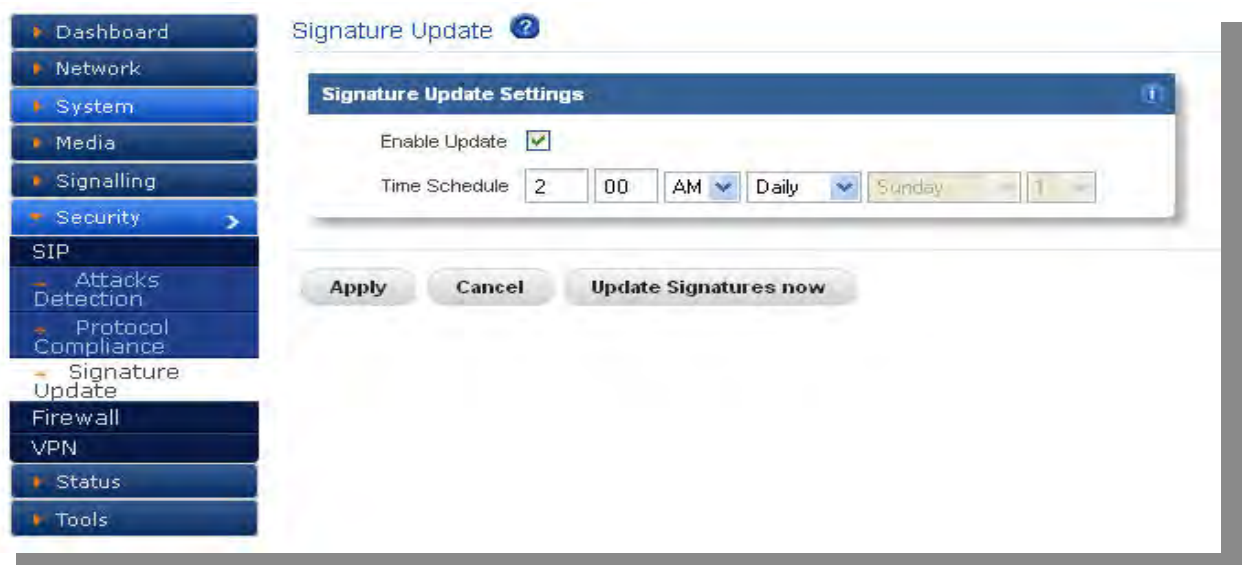


Figure 38: Signature Update

7.2 Firewall

7.2.1 Firewall Config

Navigate through **Security > Firewall > Firewall Config**

The firewall rules configuration will allow the administrator in configuring what traffic should be allowed to protected SIP PBX/Gateway network from untrusted wan zone, besides DPI enabled SIP traffic and RTP traffic.

The administrator needs to specify the source and destination networks and port numbers and protocol that will be used as the matching criteria in the filtering rule and action to be taken on matching the filtering rule.

The possible actions are to block the traffic and allow the traffic on matching the filtering rule.

The rules precedence will be in the order in which the rules configured on firewall rules table.


Shows the table with columns Name, Enabled, Src Type, Src Addr, Dst Type, Dst Addr, Protocol, Port and Action.


User can search the entries by entering the value in the Search box which appears on top right of the table.

Clicking on Add New opens a dialog with fields Name, Enabled, Src Type, Src Addr, Dst Type, Dst Addr, Protocol, Port and Action.

Single entry can be deleted by clicking on the delete button. Multiple entries can be deleted by selecting the check boxes which appears on left of each entry. Delete Selected will delete the entries which are selected.

User can sort (Ascending / Descending) the table entries by clicking on the particular column of the table for e.g. Name.

Entry can be edited by clicking on  button.

Entry can be deleted by clicking on  button.

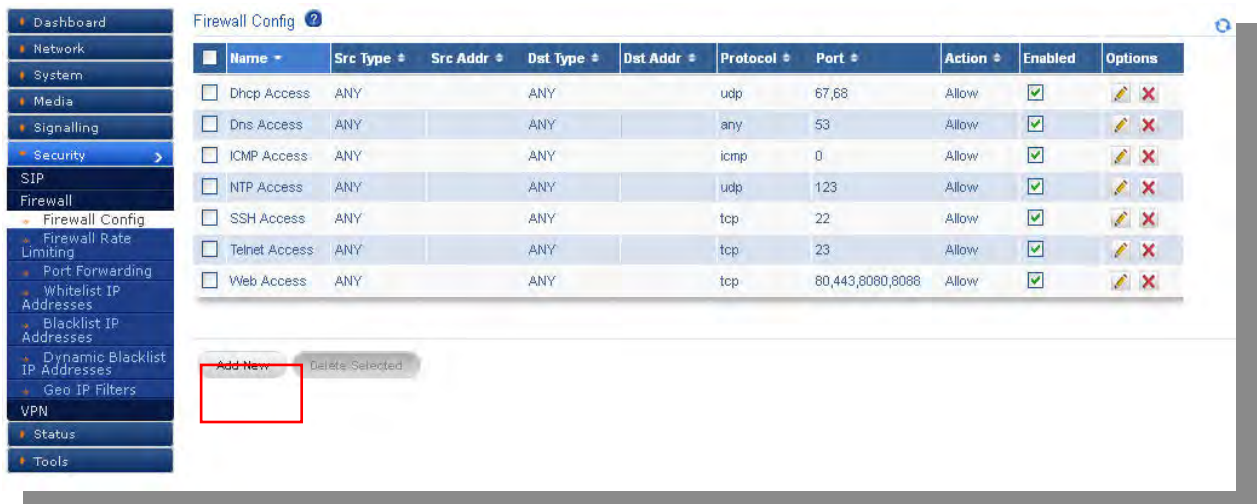


Figure 39: Firewall Configuration Results

Click Add New, to create the firewall Rule.

Create Firewall Rule

Name: NTP Access

Enabled: ☒

Src Type: ANY

Src Address:

Dst Type: ANY

Dst Address:

Protocol: any

Port:

Action: Allow

SAVE CANCEL

Figure 40: Create Firewall Rule

Name	Specify the name for the Firewall Rules for user's reference. The user can choose any name to recognize the Firewall Rules.
Enabled	It allows the user to either enable or disable Firewall Rules.
Src Type	User can select the appropriate Src type from the drop down list.
Src Address	User can configure and apply the Firewall rule to particular Source Address (Src Address). E.g.10.0.0.3
Dst Type	User can select the appropriate Dst type from the drop down list.
Dst Address	User can configure and apply the Firewall rule to particular destination

	Address (Dst Address). E.g.:192.168.0.8
Protocol	Protocols specify interactions between the communicating entities. User can select the type of protocol whether it is TCP or UDP from the drop down list.
Port	User can configure and apply the Firewall rule to particular port number.E.g.:5060
Action	User can select the action either block or action from the drop down list.

Changes can be saved by clicking on **'Save'** button and can ignore the changes by clicking on Cancel button.

7.2.2 Firewall Rate Limiting

Navigate through **Security > Firewall > Firewall Rate Limiting**

Firewall Rate Limiting allows user to configure global firewall settings.

Firewall Rate Limiting	
TCP Syn Flood Rate	1024
TCP Syn Flood Burst	128
TCP Flood Rate	4096
TCP Flood Burst	96
UDP Flood Rate	8192
UDP Flood Burst	198
ICMP Flood Rate	128
ICMP Flood Burst	64

Save Cancel

Figure 41: Firewall Rate Limiting

7.2.3 Port forwarding

Navigate through **Security > Firewall > Port forwarding**

It is used to forward incoming connection requests to internal network hosts.

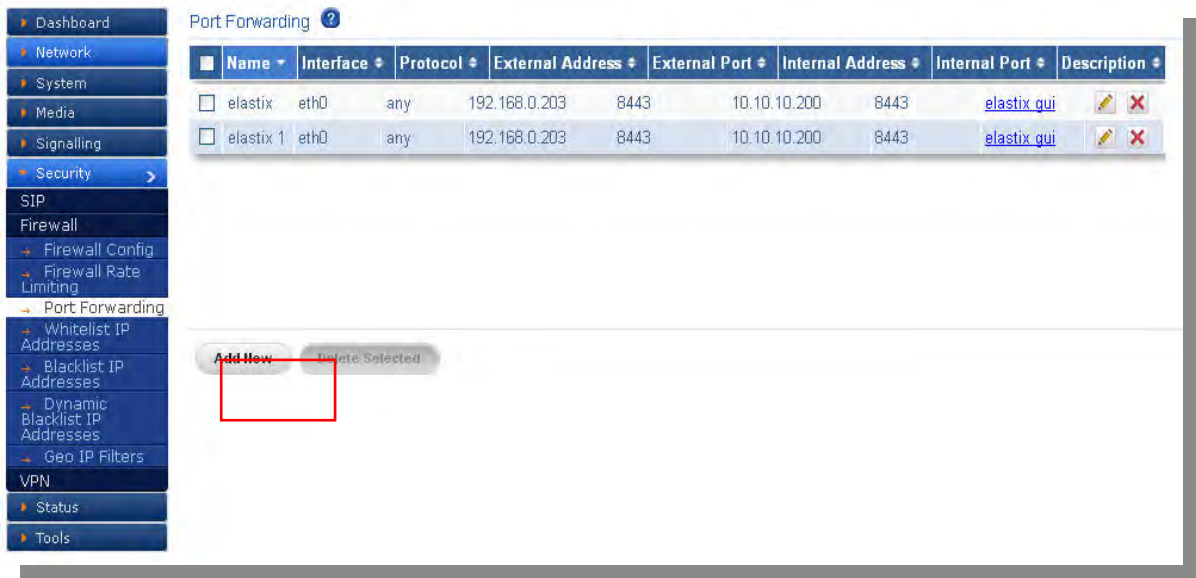


Figure 42: Port Forwarding

Click Add New, to create port forwarding Rule.

Figure 43: Create Port Forwarding Rule

Name	Specify the name for the Port forwarding for user's reference. The user can choose any name to recognize the Port forwarding.
Interfaces	Select the appropriate interfaces from the drop down list where the user desires to create Port forwarding. Ex: if user wants to create the virtual IP in wan side select Eth0, WAN Interface-192.168.10.231

	If the user wants to create the virtual IP in LAN side select Eth2,LAN Interface-10.0.0.1
Protocol	Protocols specify interactions between the communicating entities. User can select the type of protocol whether it is TCP or UDP from the drop down list.
External Address	This address assigned to you by your Internet Service Provider and allows user to enter the external address. E.g.:192.168.x.x
External Port	The port forwarding is used to identify your external address and detects open ports on your connection.
Internal Address	The internal IP address is assigned by your local network router that often begins with 192.168.x.x. It allows user to find the IP addresses in their local network.
Internal Port	It specifies the internal port that connects to the local area network (LAN).
Description	Provide the description for the Port Forwarding. (Optional)

7.2.4 White list IP Addresses

Navigate through **Security > Firewall > White list IP Addresses**

This page allows to configure the white listed IP addresses in the untrusted wan zone from which the access to communicate with the protected SIP network will be allowed by the Blox Esbc firewall.

It will also allows configuring whether the white rules take precedence over the blacklist rules (both static and dynamic) configured on the device at any instant.

White list Rules Precedes over Blacklist Rules can be saved by clicking on 'Save' button.


It shows the table with columns Name, IP Type, Address, Enabled and Comments.


User can search the entries by entering the value in the Search box which appears on top right of the table.

Clicking on Add New opens a dialog with fields Name, IP Type, Address, Enabled and Comments.

Single entry can be deleted by clicking on the delete button. Multiple entries can be deleted by selecting the check boxes which appears on left of each entry. Delete Selected will delete the entries which are selected.

User can sort (Ascending / Descending) the table entries by clicking on the particular column of the table for e.g. Name.

Entry can be edited by clicking on  button.

Entry can be deleted by clicking on  button.

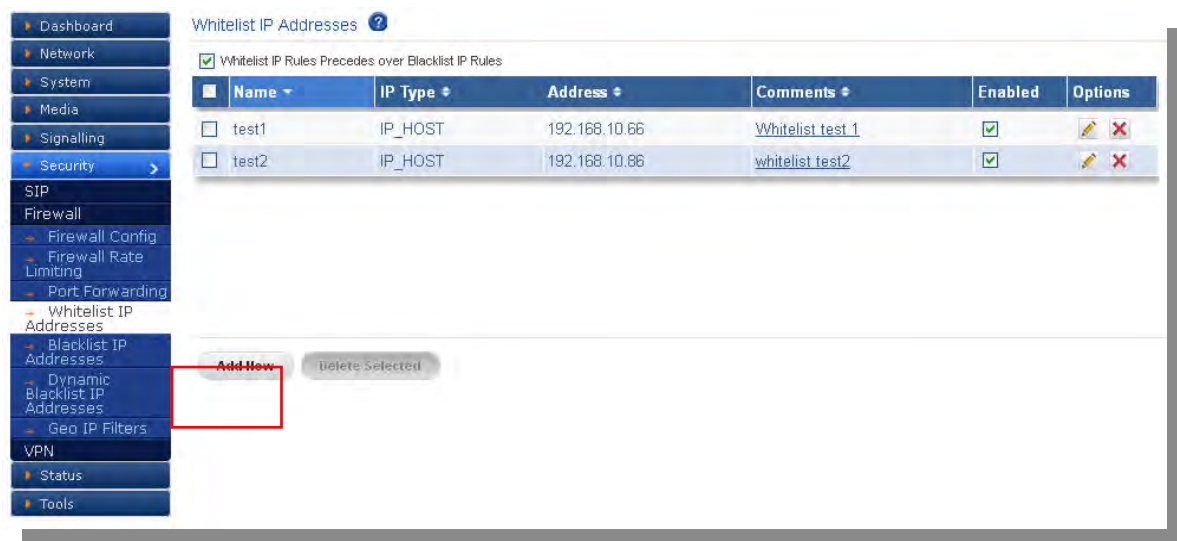
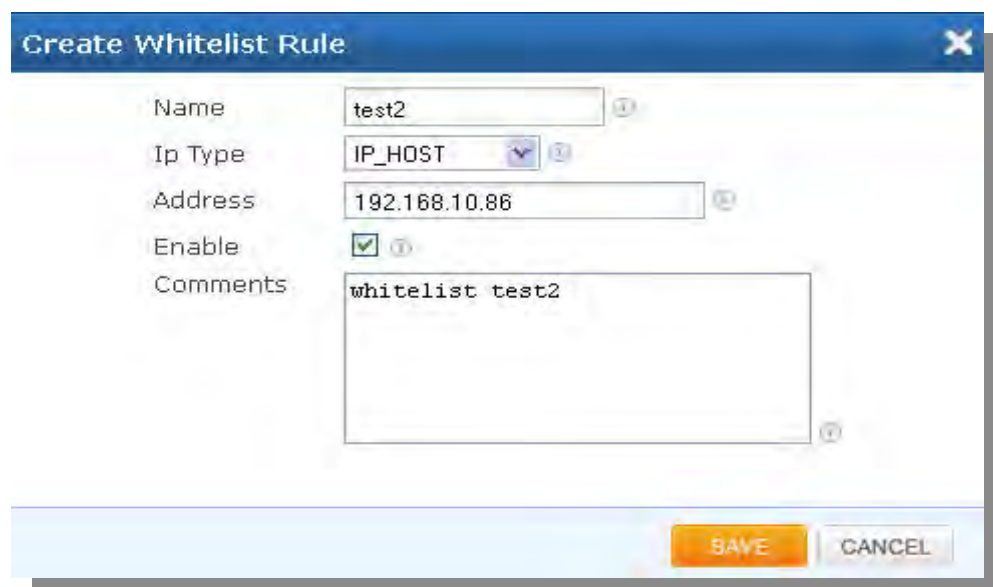


Figure 44: Whitelist IP Addresses

Click Add New, to create a Whitelist Rule.



Create Whitelist Rule

Name: test2

Ip Type: IP_HOST

Address: 192.168.10.86

Enable: ☒

Comments: whitelist test2

SAVE CANCEL

Figure 45: Create Whitelist Rule

Name	Specify the name for the White list Rules for user's reference. The user can choose any name to recognize the White list Rules.
IP Type	User can select the appropriate IP type from the drop down list. The various IP types are IP_Host, IP_Network, IP_Range, and MAC_ADDR.
Address	Specify IP Address/Netmask or IP range or MAC address.
Enable	It allows the user to either enable or disable White list Rules.
Comments	User can specify the comments in the length of 64 char's.

Changes can be saved by clicking on 'Save' button and can ignore the changes by clicking on Cancel button.

7.2.5 Blacklist IP Addresses

Navigate through **Security > Firewall > Blacklist IP Addresses**

This page allows to configure the black listed IP addresses in the untrusted wan zone from which the access to communicate with the protected SIP network will be blocked by the Blox Esbc firewall.



Figure 46: Blacklist IP Addresses

Click Add New, to create a Blacklist Rule.


Figure 47: Create Blacklist Rule


It shows the table with columns Name, IP Type, Address, Enabled and Comments.

Clicking on **Add New** opens a dialog with fields Name, IP Type, Address, Enabled and Comments.

Single entry can be deleted by clicking on the delete button. Multiple entries can be deleted by selecting the checkboxes which appears on left of each entry. **Delete Selected** will delete the entries which are selected.

User can sort (Ascending / Descending) the table entries by clicking on the particular column of the table for e.g. Name.

Entry can be edited by clicking on  button.

Entry can be deleted by clicking on  button.

Changes can be saved by clicking on '**Save**' button and can ignore the changes by clicking on **Cancel** button.

7.2.6 Dynamic Blacklist IP Addresses

Navigate through **Security > Firewall > Dynamic Blacklist IP Addresses**

The dynamic blacklist addresses are the blocking rules added by the Blox Esbc SIP deep packet inspection engine to block the traffic from attacker IP addresses for the blocking duration configured in the rules category, on detecting the attack.

The dynamic blacklist addresses page will allow the administrator to see the dynamic blacklist addresses currently configured on the device at any instant. In case if the administrator wants to override and allow the traffic from particular blacklisted IP, he can delete the address from the dynamic blacklist addresses page.

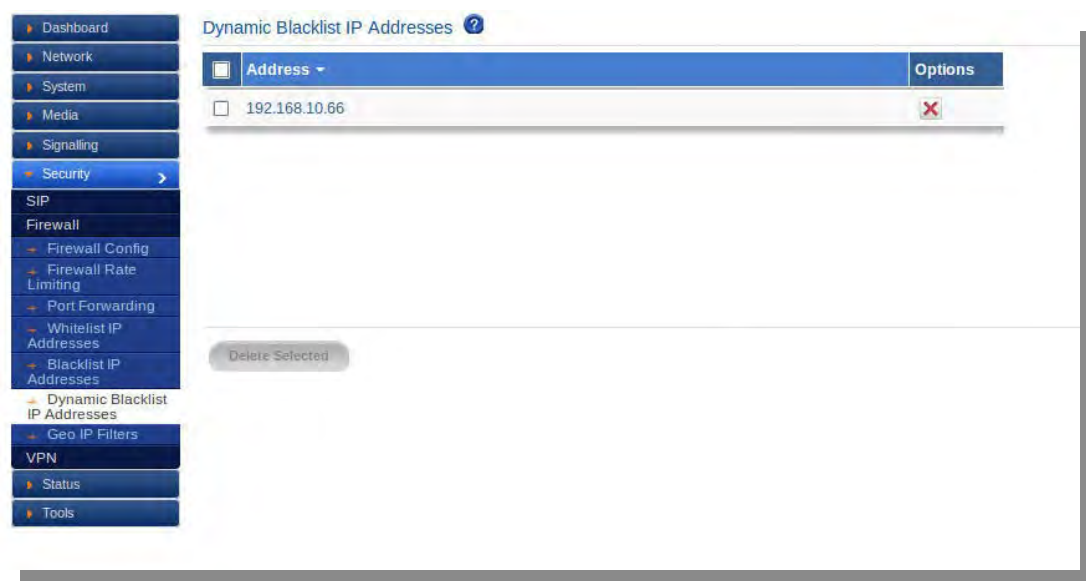


Figure 48: Dynamic Blacklist IP Addresses

It shows the table with columns Address and Options.

Single entry can be deleted by clicking on the delete button. Multiple entries can be deleted by selecting the check boxes which appears on left of each entry. “Delete Selected” will delete the entries which are selected.

User can sort (Ascending / Descending) the table entries by clicking on the particular column of the table for e.g. Name.

Entry can be deleted by clicking on button.

6.2.7 Geo IP Filters

Navigate through **Security > Firewall > Geo IP Filters**

The administrator can choose to block the traffic originating from the specific countries towards the protected SIP network, by configuring the GeoIP filter rules in Blox Esbc.


Clicking on Allow All Countries will allow all the countries and Block All Countries will block all the countries.

Clicking on Update Geo IP will download the latest database from website and replace the existing country database.

It shows the table with columns Country Name and Allowed.

User can search the entries by entering the value in the Search box which appears on top right of the table.

User can sort (Ascending / Descending) the table entries by clicking on the particular column of the table for e.g. Name.

Entry can be edited by clicking on  button.

Changes can be saved by clicking on 'Save' button and can ignore the changes by clicking on Cancel button.



Country Name	Allowed	Options
RUSSIAN FEDERATION	<input checked="" type="checkbox"/>	
SYRIAN ARAB REPUBLIC	<input checked="" type="checkbox"/>	
SUDAN	<input checked="" type="checkbox"/>	
NIGERIA	<input checked="" type="checkbox"/>	
KOREA, REPUBLIC OF	<input checked="" type="checkbox"/>	
CHINA	<input checked="" type="checkbox"/>	
UKRAINE	<input checked="" type="checkbox"/>	
ALGERIA	<input checked="" type="checkbox"/>	

Figure 49: Geo IP Filters

7.3 VPN

A virtual private network (VPN) tunnel provides a secure communication channel either between two gateway VPN firewalls or a remote VPN client and gateway VPN firewall. As a result, the IP address at least one of the tunnel endpoints needs to be known in advance in order for the other tunnel endpoint to establish (or reestablish) the VPN tunnel.

7.3.1 IPsec VPN

Navigate through **VPN > IPsec VPN**

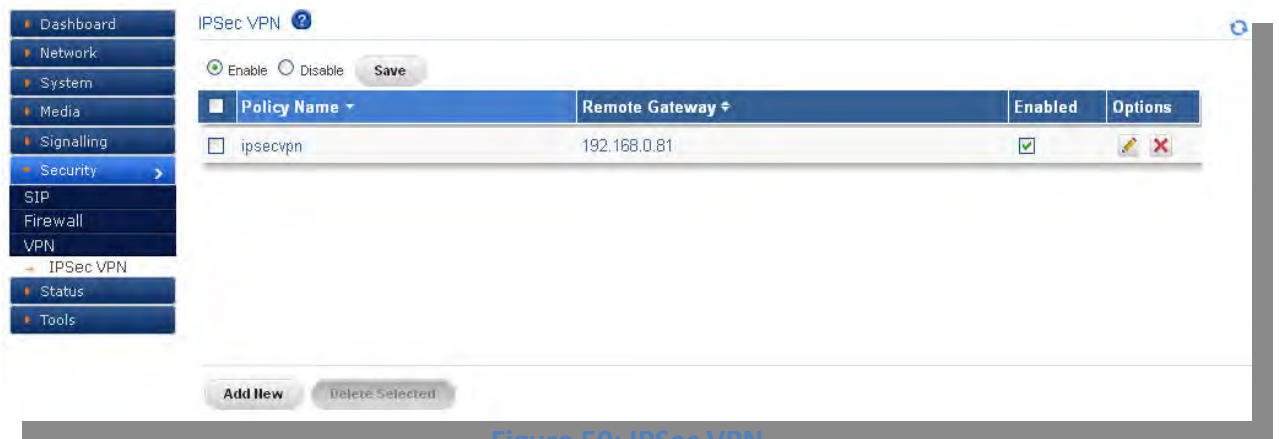


Figure 50: IPsec VPN

Policy Settings

IPsec has the following two modes of forwarding data across a network:

- Tunnel mode
- Transport mode

Each differs in its application as well as the amount of overhead added to the passenger packet. These modes are described in more detail in the next two sections.

Tunnel Mode

It works by encapsulating and protecting an entire IP packet. Because tunnel mode encapsulates or hides the IP header of the pre-encrypted packet, a new IP header is added so that the packet can be successfully forwarded. The encrypting devices themselves own the IP addresses used in this new header.

It can be configured with either or both IPsec protocols (ESP and AH). Tunnel mode results in additional packet expansion of approximately 20 bytes because of the new IP header.

Tunnel mode is widely considered more secure and flexible than transport mode. IPsec tunnel mode encrypts the source and destination IP addresses of the original packet, and hides that information from the unprotected network.

Create IPSec VPN Rule

Policy Settings | IKE | IPSec | Advanced

Enable ☒

Policy Name ⓘ

Mode ☒ Tunnel ☐ Transport

Policy Type ☒ P2P ☐ Road Warrior

Local Gateway

Local Network

Remote Gateway

Remote Network

SAVE **CANCEL**

Figure 51: Policy Settings

Enable	This allows the user to either enable or disable policy settings. If it's enabled, then this policy is deployed.
Policy Name	Enter the policy name for the IPSec VPN for user reference.
Mode	User can select different modes p2p / Road warrior depending on these 2, tunnels and transport can be selected.
Policy Type	User can select either p2p / Road warrior policy type.
Local gateway	It specifies the gateway IP of the device. E.g.: 192.168.x.x
Local network	Network behind the gateway need to be accessed. Eg: 192.168.0.0/24
Remote gateway	Enter the Remote gateway IP. E.g.: 192.168.x.x
Remote network	It specifies Remote gateway to be accessed. Eg: 192.168.1.0/24

IKE

To implement a VPN solution with encryption, periodic changing of session encryption keys is necessary. Failure to change these keys makes the VPN susceptible to brute force decryption attacks. IPSec solves the problem with the IKE protocol, which makes use of two other protocols to

authenticate a crypto peer and to generate keys. IKE uses a mathematical algorithm called a Diffie-Hellman exchange to generate symmetrical session keys to be used by two crypto peers.

IKE also manages the negotiation of other security parameters such as the data to be protected, the strength of the keys, the hash methods used, and whether the packets are protected from anti-replay. ISAKMP normally uses UDP port 500 as both the source and destination port.

Figure 52: IKE

IKE Exchange Mode	User can select two modes like Main or aggressive mode to be sustained.
Lifetime	It specifies time after the renegotiation of phase 1 happens
Encryption Algorithm	It can be used during phase 1 negotiation
Hash Algorithm	It is mainly used for authentication in phase 1.
Authentication Method	It allows a single method (PreSharedKey) to authenticate the IKE mode instance of all methods. The selected authentication allows user to configure the field.
Preshared Key	This secret key is mainly used for authentication.
DH Group	Key exchange protocol allows two parties without any initial shared sheet to create one securely.

IPSec

Transport: can use AH/ESP mode.

AH (Authentication Header)

The AH protocol (IP protocol 51) forms the other part of IPSec. It does not encrypt data in the usual sense, by hiding the data but it adds a tamper-evident seal to the data. It also protects the non-mutable fields in the IP header carrying the data, which includes the address fields of the IP header.

The AH protocol should not be used alone when there is a requirement for data confidentiality.

ESP (Encapsulating Security Protocol)

The ESP header (IP protocol 50) forms the core of the IPSec protocol. This protocol, in conjunction with an agreed-upon set of security Parameters or transform set, protects data by rendering it indecipherable. This protocol encrypts the data portion of the packet only and uses other protections (HMAC) for other protections (data integrity, anti-replay, and man-in-the-middle). Optionally, it can also provide for authentication of the protected data.



Figure 53: IPSec Settings

Transport	User can use AH/ESP mode
Lifetime	It specifies the time, after the renegotiation of phase 2 happens.
PFS Group	Perfect Forward Secrecy of keys does not compromise keys.

Encryption Algorithm	It can be used during phase 2 negotiations. And allows user to select different algorithms from the dropdown list.
Authentication Algorithm	It is used for authentication in phase 2. It allows user to select anyone of the algorithms from the dropdown list.

Advanced

The screenshot shows the 'Create IPSec VPN Rule' dialog box with the 'Advanced' tab selected. The settings are as follows:

- Enable compression: on
- NAT Traversal: off
- Enable Dead Peer Detection: ☒
- DPD Delay: 20 in seconds
- DPD Maxfail: 30 in seconds

Buttons: SAVE, CANCEL

Figure 54: Advanced

Enable compression	User can used deflate alg to compress traffic.
Nat Traversal	If the gate is noted then this option has to be on or force else select no.
Enable Dead Peer Detection	User can enable keep alive signals for the connection.
DPD Delay	It allows keep alive signals that are sent for this connection.
DPD Max fail	It specifies Max number of second keep alive signals that are sent for connection.

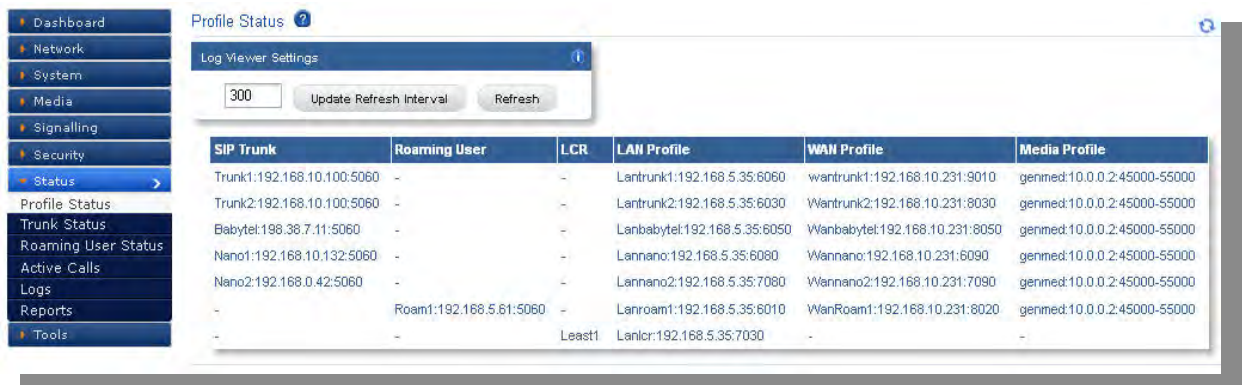
8. Status

8.1 Profile Status

Navigate through Status > Profile Status

Profile Status shows the configured SIP trunks, roaming users, least counting routing, along with the corresponding LAN profile, WAN profile and Media profile IP addresses and port numbers.

And also it shows the Log viewer settings, which allows user to refresh the page and edit the refresh time interval.



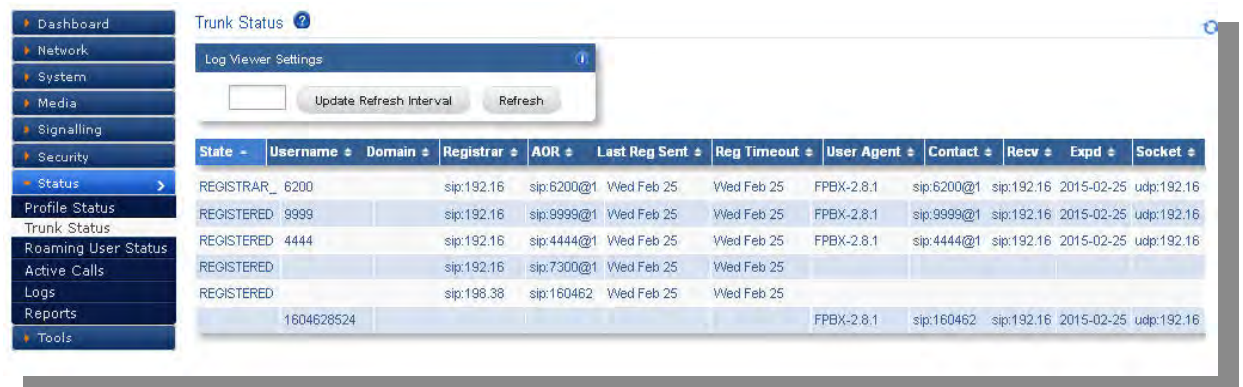
SIP Trunk	Roaming User	LCR	LAN Profile	WAN Profile	Media Profile
Trunk1:192.168.10.100:5060	-	-	Lantrunk1:192.168.5.35:6060	wantrunk1:192.168.10.231:9010	genmed:10.0.0.2:45000-55000
Trunk2:192.168.10.100:5060	-	-	Lantrunk2:192.168.5.35:6030	Wantrunk2:192.168.10.231:8030	genmed:10.0.0.2:45000-55000
Babytel:198.38.7.11:5060	-	-	Lanbabytel:192.168.5.35:6050	Wanbabytel:192.168.10.231:8050	genmed:10.0.0.2:45000-55000
Nano1:192.168.10.132:5060	-	-	Lannano:192.168.5.35:6080	Wannano:192.168.10.231:6090	genmed:10.0.0.2:45000-55000
Nano2:192.168.0.42:5060	-	-	Lannano2:192.168.5.35:7080	Wannano2:192.168.10.231:7090	genmed:10.0.0.2:45000-55000
-	Roam1:192.168.5.61:5060	-	Lanroam1:192.168.5.35:6010	WanRoam1:192.168.10.231:8020	genmed:10.0.0.2:45000-55000
-	-	Least1	Lanlcr:192.168.5.35:7030	-	-

Figure 55: Profile Status

8.2 Trunk Status

Navigate through Status > Trunk Status

Trunk Status shows the current status of the configured trunks in the Blox Esbc. It contains user name, domain, registrar etc. And also it shows the Log viewer settings which allows user to refresh the page and edit the refresh time interval.



State	Username	Domain	Registrar	AOR	Last Reg Sent	Reg Timeout	User Agent	Contact	Recv	Expd	Socket
REGISTRAR_	6200		sip:192.16	sip:6200@1	Wed Feb 25	Wed Feb 25	FPBX-2.8.1	sip:6200@1	sip:192.16	2015-02-25	udp:192.16
REGISTERED	9999		sip:192.16	sip:9999@1	Wed Feb 25	Wed Feb 25	FPBX-2.8.1	sip:9999@1	sip:192.16	2015-02-25	udp:192.16
REGISTERED	4444		sip:192.16	sip:4444@1	Wed Feb 25	Wed Feb 25	FPBX-2.8.1	sip:4444@1	sip:192.16	2015-02-25	udp:192.16
REGISTERED			sip:192.16	sip:7300@1	Wed Feb 25	Wed Feb 25					
REGISTERED			sip:198.38	sip:160462	Wed Feb 25	Wed Feb 25					
	1604628524						FPBX-2.8.1	sip:160462	sip:192.16	2015-02-25	udp:192.16

Figure 56: Trunk Status

8.3 Roaming User Status

Navigate through Status > Roaming User Status

It displays the current status of roaming users configured in Blox Esbc. It contains user name, domain, registrar etc. And also it shows the Log viewer settings which allows user to refresh the page and edit the refresh time interval.

Username	Domain	User Agent	Contact	Received	Expires	Last Modified	Socket	Attr
1010		Yealink SIP-T22P 7.72.0.30	sip:1010@192.168.10.69:5064	sip:192.16	2015-02-25	2015-02-25	udp:192.16	udp:192.16
1004		Yealink SIP-T22P 7.72.0.75	sip:1004@192.168.10.52:5062	sip:192.16	2015-02-25	2015-02-25	udp:192.16	udp:192.16

Figure 57: Roaming User Status

8.4 Active calls

Navigate through **Status > Active calls**

It displays status of the live calls along with Dialing ID, Call ID, From URI, Caller contact, To URI, Callee Contact, start time, timeout and profiles etc.

And also it shows the Log viewer settings, which allows user to Update, refresh interval and refresh the page.

Dialling ID	Call ID	From URI	Caller Contact	Caller Sock
16184550446632	6f3e5862140a4b8025a665a413dd06ff@192.168.10.100:5060	sip:6002@192.168.10.100	sip:6002@192.168.10.100:5060	udp:192

Figure 58: Active Calls

8.5 Logs

8.5.1 Signaling Logs

Navigate through **Status> Logs> Signaling Logs**

Signaling logs demonstrates complete logs of the SIP request methods received by the Blox Esbc.

The Log viewer settings allows user to update refresh interval and Refresh the Log viewer settings.

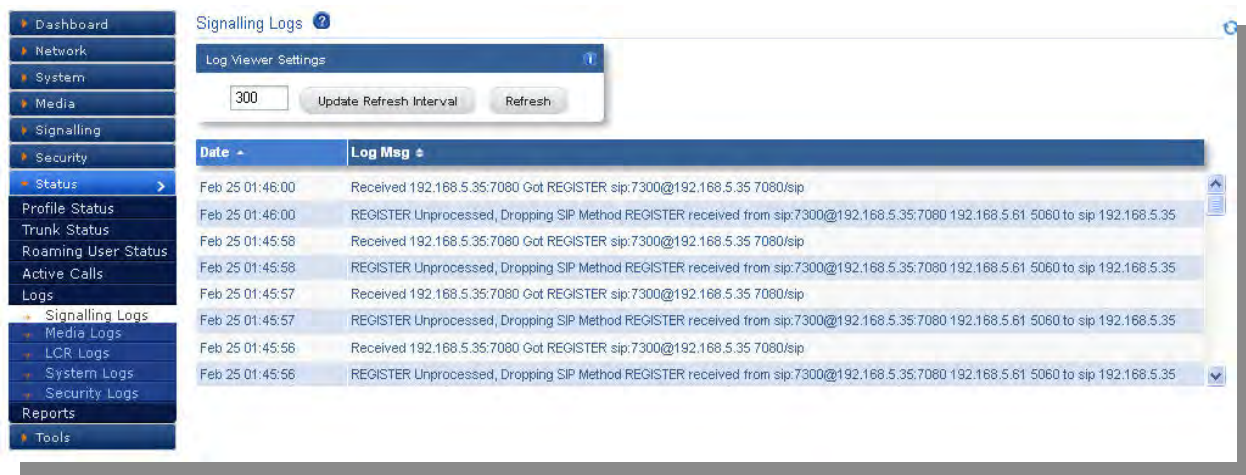


Figure 59: Signaling Logs

8.5.2 Media Logs

Navigate through **Status> Logs> Media Logs**

It shows the log messages about the media which are sending and received by the Blox Esbc. The Log viewer settings allows user to update refresh interval and Refresh the Log viewer settings.

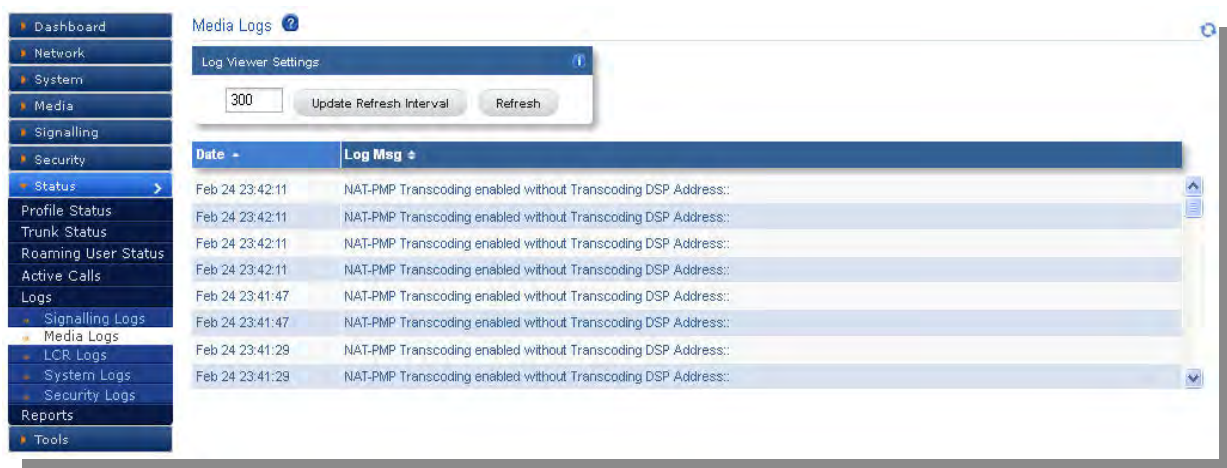


Figure 60: Media Logs

8.5.3 LCR Logs

Navigate through **Status> Logs> LCR Logs**

It displays call logs which are made through Least Cost Routing. The Log viewer settings allows user to update refresh interval and Refresh the Log viewer settings.

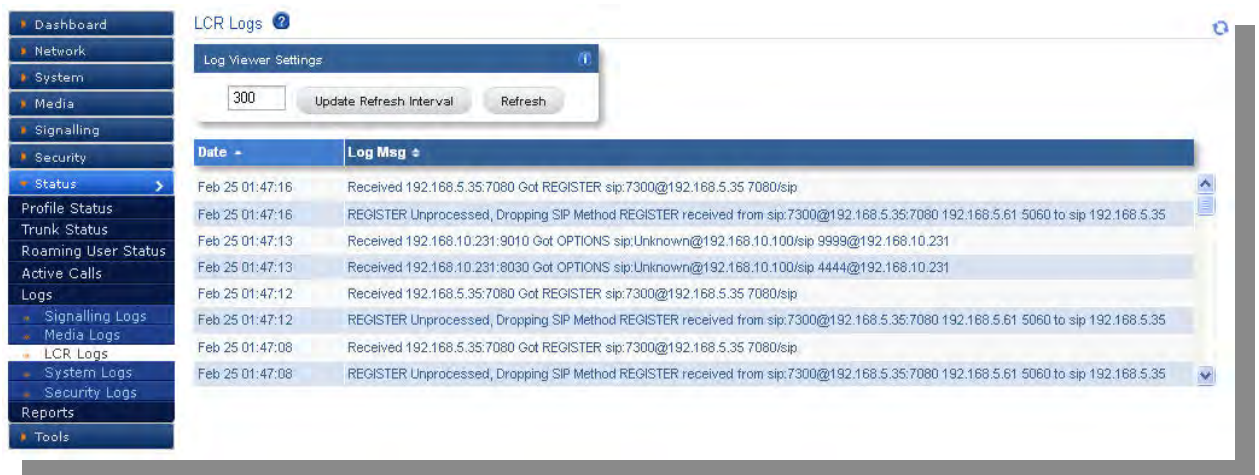


Figure 61: LCR Logs

8.5.4 System Logs

Navigate through **Status > Logs > System logs**

System log shows all the log messages of Blox Esbc. The Log viewer settings allows user to update refresh interval and Refresh the Log viewer settings.

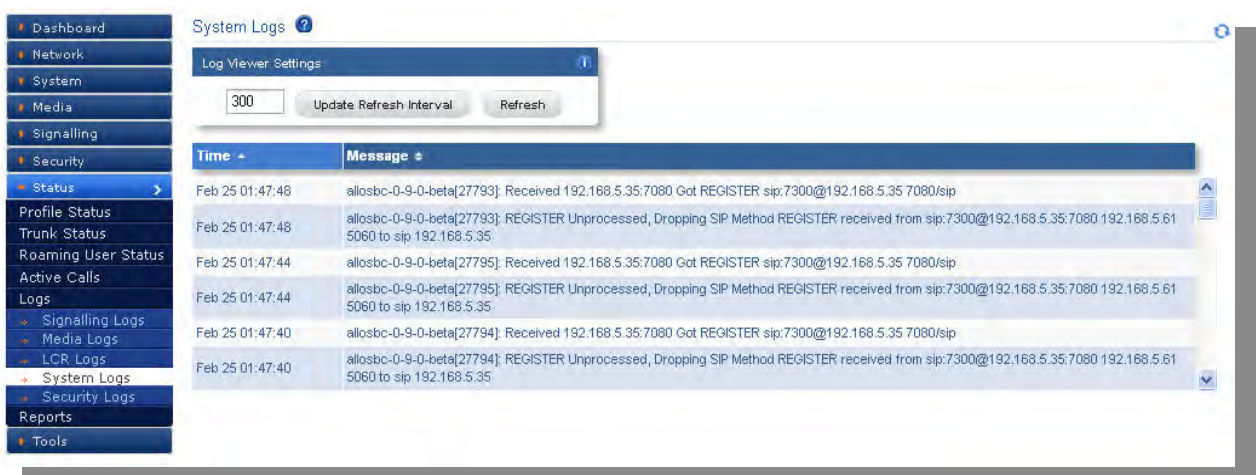


Figure 62: System Logs

8.5.5 Security Logs

Navigate through **Status > Logs > Security logs**

A security log provides a track security related information in Blox Esbc with Signature ID, Signature category and name. It also shows the Time stamp information, Source IP & Port, Destination IP & Port and type of protocol whether it is TCP or UDP.

The Log viewer settings allows user to update refresh interval and Refresh the Log viewer settings.

Dashboard	Security Logs ?
Network	Log Viewer Settings
System	300 Update Refresh Interval Refresh
Media	
Signalling	
Security	
Status	
Profile Status	
Trunk Status	
Roaming User Status	
Active Calls	
Logs	
Signalling Logs	
Media Logs	
LCR Logs	
System Logs	
Security Logs	
Reports	
Tools	

Time	ID	Category	Category Name	Message	Src IP	Src Port	Dst IP	Dst Port	Protocol
12/04-16:23:55	70110001	7011	SIP Extensions Discovery	"Sig: SIP Extensions Identification Attempt"	192.168.5.61	5060	192.168.5.35	6060	UDP Blacklist
12/04-16:23:55	70110001	7011	SIP Extensions Discovery	"Sig: SIP Extensions Identification Attempt"	192.168.5.61	5060	192.168.5.35	6030	UDP Blacklist
12/04-16:23:51	70110001	7011	SIP Extensions	"Sig: SIP Extensions Identification"	192.168.5.61	5060	192.168.5.35	6060	UDP Blacklist

Figure 63: Security Logs

9. Reports

9.1 CDR Reports

Navigate through **Status > Reports > CDR Reports**

Call Detailed Reports (CDR) displays detailed information about the calls through Blox Esbc.

Dashboard	CDR Reports ?
Network	Log Viewer Settings
System	300 Update Refresh Interval Refresh
Media	
Signalling	
Security	
Status	
Profile Status	
Trunk Status	
Roaming User Status	
Active Calls	
Logs	
Reports	
CDR Reports	
Tools	

ID	Time	Method	Source	Channel	Destination	Dest. Channel	SIP Code	SIP Reason	Duration
1587	23:42:11 2015-02-24	INVITE	sip:100@19	sip:192.16	sip:7300@1	487	Request Te	0	0
1586	23:41:29 2015-02-24	INVITE	sip:100@19	sip:192.16	sip:7300@1	487	Request Te	0	0
1585	23:32:30 2015-02-24	INVITE	sip:100@19	sip:192.16	sip:7300@1	487	Request Te	0	0
1584	23:16:55 2015-02-24	INVITE	sip:7300@1	sip:192.16	sip:100@19	sip:7300@1	200	OK	1 8
1582	23:15:20 2015-02-24	INVITE	sip:7300@1	sip:192.16	sip:100@19	sip:7300@1	200	OK	1 8
1580	23:13:15 2015-02-24	INVITE	sip:7300@1	sip:192.16	sip:100@19	sip:7300@1	200	OK	1 23
1578	23:12:10 2015-02-24	INVITE	sip:7300@1	sip:192.16	sip:100@19	sip:7300@1	200	OK	1 8
1576	23:02:56 2015-02-24	INVITE	sip:7300@1	sip:192.16	sip:100@19	sip:7300@1	200	OK	1 9

Figure 64: CDR Reports

10. Tools

10.1 Administration

Navigate through **Tools > Administration**

User can do factory reset by clicking on Factory Reset button.

They restart Blox Esbc services by clicking on Restart STM Services button.

User can reboot device by clicking on Reboot button.

User can shutdown device by clicking on Shutdown button.

User can take back up of the configuration by clicking on Config Back-Up button.

Restoring the configuration can be done by selecting the configuration file from the system and clicking on the Config Restore button which reboots the machine on success.

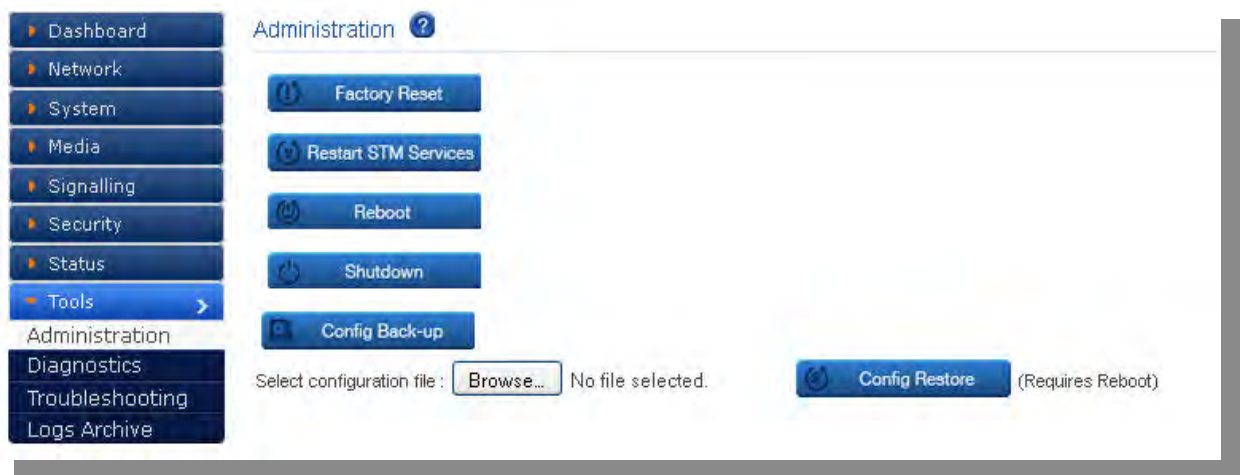


Figure 65: Administration

10.2 Diagnostics

Navigate through **Tools > Diagnostics > Run Diagnostics**

10.2.1 Run Diagnostics

User can run diagnostics by clicking on Run Diagnostics button and result can be seen in the text region.

Diagnostics report can be downloaded by clicking on the Get Report button

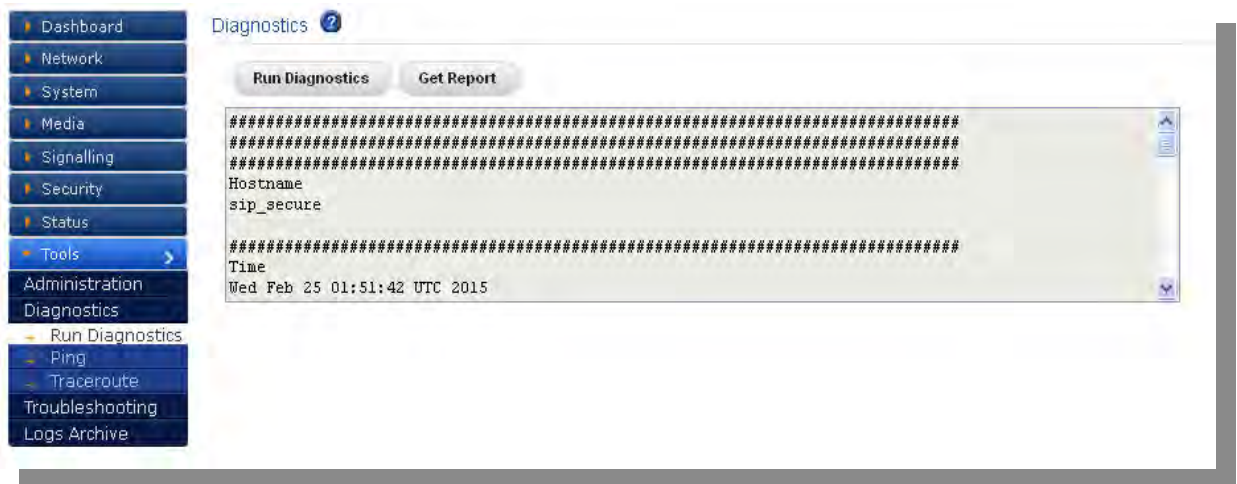


Figure 66: Diagnostics

10.2.2 Ping

Navigate through **Tools > Diagnostics > Ping**

User can ping a host by entering values for host IP / Domain Name and selecting the count from the list.

Ping button will send a ping request to the host and Reset button clears the entered values.

Ping result is shown in the text area which appears below the ping and reset buttons.

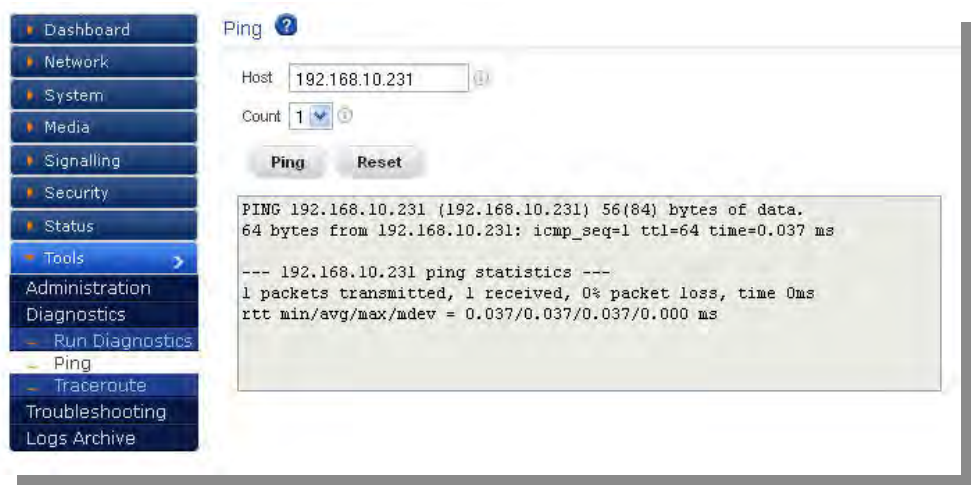


Figure 67: Ping Result

10.2.3 Trace route

Navigate through **Tools > Diagnostics > Traceroute**

User can trace route a host by entering values for host IP / Domain Name, hop count and enabling the ICMP by clicking ICMP checkbox.

Traceroute button will send a trace route request to the host and Reset button clears the entered values.

Traceroute result is shown in the text area which appears below the ping and reset buttons.

The screenshot shows the 'Traceroute' section of a network management interface. On the left is a sidebar menu with options: Dashboard, Network, System, Media, Signalling, Security, Status, Tools (highlighted with a right arrow), Administration, Diagnostics, Run Diagnostics, Ping, Traceroute, Troubleshooting, and Logs Archive. The main area is titled 'Traceroute' with a help icon. It contains input fields for 'Host' (192.168.10.231) and 'Hop Count' (3), and an unchecked 'ICMP' checkbox. Below these are 'Traceroute' and 'Reset' buttons. A text box displays the results: 'traceroute to 192.168.10.231 (192.168.10.231), 3 hops max, 60 byte packets' followed by a table of hop data.

Hop	IP Address	RTT (ms)
1	192.168.10.231 (192.168.10.231)	0.036 ms 0.007 ms 0.007 ms

Figure 68: Trace Route Result

10.3 Trouble shooting

Navigate through **Tools > Trouble Shooting**

By Clicking on Enable DPI or Disable DPI button which enables or disables DPI.

The screenshot shows the 'Troubleshooting' section of the same network management interface. The sidebar menu is identical, with 'Tools' highlighted. The main area is titled 'Troubleshooting' with a help icon. It features a single button labeled 'Enable DPI'.

Figure 69: Trouble Shooting

10.4 Logs Archive

Navigate through **Tools > Logs Archive**

After the device storage has reached its limit, logs are stored in USB storage device if one is connected. Logs archive summary is listed in the text area.



Figure 70: Logs Archive

The image features a solid blue background. In the top right corner, there is a curved, multi-colored line that transitions from yellow to orange to red. A similar curved, multi-colored line is located in the bottom left corner, also transitioning from yellow to orange to red. The text "Thank You" is centered in the middle of the image in a white, sans-serif font.

Thank You