

Grandstream Networks, Inc.

GXW45xx Series

User Manual



GXW4500 Series - User Manual

Thank you for purchasing the Grandstream GXW450X Digital VoIP Gateway. The GXW450X offers an easy-to-manage, easy-to-configure IP communications solution for any business with virtual and/or branch locations. The GXW450X supports popular voice codecs and is designed for compatibility and interoperability with third-party SIP providers, thus enabling you to fully leverage the benefits of VoIP technology, integrate an I system into a VoIP network, and efficiently manage GXW450x supports SNMP (Simple Network Management Protocol) which is widely used in net management for network monitoring for collecting information about monitored devices. To configure SNMP settings, go to GXW450x Web GUI—System Settings—SNMPcommunication costs.

This manual will help you learn how to operate and manage your GXW450X Digital Gateway and make the best use of its many upgraded features including simple and quick installation, multi-party conferencing, and direct IP-IP Calling. This Digital VoIP Gateway is very easy to manage and scapecifically designed to be an easy-to-use and affordable VoIP solution for large and medium-sized enterprises

Safety Compliances

The GXW450X is compliant with various safety standards including FCC/CE. Its power adapter is compliant with UL standards.

Warning:

Use only the power adapter included in the GXW450X package. The use of an alternative power adapter may permanently damage the unit.

Warranty:

Grandstream has a reseller agreement with our reseller customers. End users should contact the company from whom the product was purchased, for replacement, repair, or refund.

If you purchased the product directly from Grandstream, contact your Grandstream Support for an RMA (Return Materials Authorization) number. Grandstream reserves the right to change the warranty policy without prior notification.

Caution

Changes or modifications to this product not expressly approved by Grandstream, or operation of this product in any way other than as detailed by this U Manual, could void your manufacturer warranty.

GATEWAY GXW450X OVERVIEW

The GXW450X series are E1/T1/J1 Digital VoIP Gateways that allow digital PSTN and ISDN trunks to be integrated with VoIP networks. By connecti GXW450X series with a VoIP network and traditional PBX or E1/T1/J1 providers, businesses can drastically increase the number of PSTN/ISDN trun integrated with their VoIP network and the concurrent calls supported. The GXW450X series offers three models that provide 1, 2, or 4 T1/E1/J1 sp and support 30, 60, or 120 concurrent calls.

Feature Highlights

The following table contains the major features of the GXW450X:

GXW450X

- o 1,2 or 4 Software configurable E1/T1/J1 ports
- o Support of PRI, SS7and MFC R2 Signaling protocols
- o Dual Gigabit Auto-sensing RJ45 Network ports with integrated NAT router
- Support of T.38 FAX for creating Fax-over-IP
- o Support of a wide range of voice codecs, including G.722, G.729, iLBC, OPUS, and more
- o TLS and SRTP security encryption technology to protect calls and accounts
- o Support of multi-language voice prompt
- o Supports up to 120 concurrent calls

GXW450X Technical Specifications

The following table resumes all the technical specifications including the protocols/standards supported, voice codecs, languages, and upgrade/provisioning settings for the GXW450X

Interfaces						
T1/E1/J1 Interface	1/2/4 RJ45 ports, supporting up to 30/60/120 simultaneous VoIP calls					
Network Interfaces	Dual self-adaptive Gigabit ports (switched or routed)					
Peripheral Ports	(2) USB 3.0, (1) SD card interface					
LED Indicators	WAN, LAN, T1/E1/J1					
LCD Display	28×32 dot matrix graphic LCD with DOWN and OK buttons					
Reset Switch	Yes, long press for factory reset and short press for the reboot.					
Voice Capabilities						
Voice-over-Packet Capabilities	LEC with NLP Packetized Voice Protocol Unit, 128ms-tail-length carrier grade Line Echo Cancellation, Dynamic Jitter Bu Modem detection & auto-switch to G.711					
Voice and Fax Codecs	G.711 A-law/U-law, G.722, G.723.1 5.3K/6.3K, G.726, G.729A/B, Opus, iLBC, GSM-FR, AAL2-G.726-32					
Fax over IP	38 compliant Group 3 Fax Relay up to 14.4kpbs and auto-switch to G.711 for Fax Passthrough, Fax data pump V.17, V 27ter, V.29 for T.38 fax relay.					
Voice-quality Enhancement	Echo cancellation (G.168-2004), Jitter buffer, Silence suppression (VAD, CNG), PLC					
QoS	Layer 2 QoS (802.1Q, 802.1p) and Layer 3 (ToS, DiffServ, MPLS) QoS					
Signaling & Control						
DTMF Methods	In-band audio, RFC2833, and/or SIP INFO					
	SIP (RFC 3261) over UDP/TCP/TLS, PRI, SS7, MFC R2, RBS (pending)					
Divital Cinnelling	PRI switch types: Euro ISDN, nation, Q.SIG					
Digital Signaling	CAS: MFC R2 (Argentina, Brazil, China, Czech Republic, Colombia, Ecuador, Indonesia, ITU, Mexico, Philippines, Venezue					
	SS7: ITU, ANSI, China					
Upgrade	Firmware upgrade via TFTP / HTTPS or local HTTP upload					
Device Management	Syslog, HTTPS, Web browser, voice prompt, backup and restore, port capture, and packet capture					
Network Protocols	TCP/UDP/IP, RTP/RTCP, ICMP, ARP, DNS, DDNS, DHCP, NTP, TFTP, SSH, HTTP/HTTPS, PPPoE, STUN, SRTP, TLS, LDAP, PI Frame Relay (pending), IPV6, OpenVPN®					
Status and statistic	Call status and history, device status monitoring, and ISDN status monitoring					
Security						
Media Encryption	SRTP, TLS, HTTPS, SSH, 802.1X					
User-defined ports	SIP port, RTP port, HTTP/HTTPS port					

Advanced Defense	Fail2ban, alert events, Whitelist, Blacklist, strong password-based access control					
Physical						
Universal Power	Input: 100-240VAC, 50/60Hz					
Supply	Output: DC+12V/2A					
	GXW4501: Unit Weight: 2350g; Package Weight: 3130g					
Physical &	GXW4502: Unit Weight: 2360g; Package Weight: 3140g					
Dimensions	GXW4504: Unit Weight: 2380g; Package Weight: 3160g					
	Unit Dimensions: 485mm(L) x 191mm(W) x 46.2mm (H)					
Temperature and	Operating: 32 – 113°F / 0 ~ 45°C, Humidity 10 – 90% (non-condensing)					
Humidity	Storage: 14 – 140°F / -10 ~ 60°C, Humidity 10 – 90% (non-condensing)					
Mounting	Rack mount & Desktop					
Additional Features						
	Web UI: English, Simplified Chinese, Traditional Chinese, Spanish, French, Portuguese, German, Russian, Italian, Polish, C					
Multi-Language Support	Customizable IVR/voice prompts: English, Chinese, British English, German, Spanish, Greek, French, Italian, Dutch, Polish Portuguese, Russian, Swedish, Turkish, Hebrew, Arabic;					
	Customizable language pack to support any other languages					
	FCC: 47 C.F.R FCC Part 15 Class B; 47 C.F.R FCC Part 68 (TIA-968-B Section 5.2.4 (T1+ISDN))					
Compliance	CE: EN 55032,EN 55035,EN 61000-3-2,EN 61000-3-3,EN 60950-1,TBR 4 (E1+ISDN),TBR 12 (E1),TBR 13 (E1+ISDN)					
	RCM: AS/NZS CISPR 32,AS/NZS 61000.3.2,AS/NZS 61000.3.3,AS/NZS 60950.1,AS/ACIF S016(E1),AS/ACIF S038(E1+ISDN					
	Other: ITU K.21 (Enhanced Levels); UL 60950-1 (Power adapter)					

Table 2: GXW450X Technical Specifications

GETTING STARTED

This chapter provides basic installation instructions including the list of the packaging contents and also information for obtaining the best perfor with the GXW450X.

Equipment Packaging

Unpack and check all accessories. Equipment includes

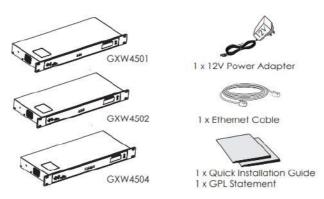


Figure 1: GXW450X Package Contents

Connecting the GXW450X

Connecting the GXW450X gateway is easy. Follow these steps to connect your GXW450X gateway to the Internet and access the unit's configurati pages.

- 1. Connect one end of a straight-through RJ45 Ethernet cable into the WAN port of the GXW450X; connect the other end to the uplink port of a Ethernet switch/hub.
- 2. Connect the 12V DC power adapter to the DC 12V power jack on the back of the GXW450X. Insert the main plug of the power adapter into a protected power outlet.
- 3. Connect one end of the T1/E1/J1 cable provided by the service provider into the T1/E1/J1 port of the GXW450X; connect the other end to the T1/E1/J1 wall jack.
- 4. Wait for the GXW450X to boot up. The front LCD display will show the GXW450X hardware information when the boot process is completed.
- 5. Once the GXW450X is successfully connected to the network via the WAN port, the Network LED indicator will be lit green, and an IP address shown on the LCD display.

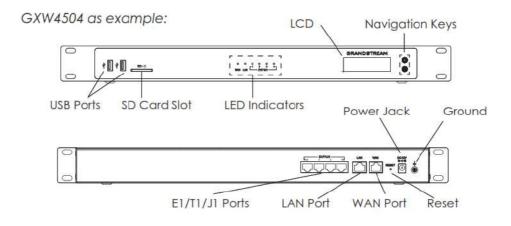


Figure: 2 Diagram of GXW4504 Back and Front Panel

WAN/LAN ports	Ethernet ports used to connect the GXW to the local and external network
RESET	Factory Reset button. Press and hold for a while to reset the factory default settings.
Power Jack	Power adapter connection
E1/T1/J1 ports	Digital port to be connected to a digital line.
USB port	2 Ports used to connect external USB drives to the GXW
SD Card Slot	Reads the SD cards memory
Ground	The ground screw needs to be connected to the ground.

Table 3: Definitions of the GXW450X Connectors

Using GXW450X Keypad Menu

The keypad menu of the GXW450X consists of 2 buttons: OK and Down keys to navigate different options.

- 1. Press the "OK" key to start browsing menu options.
- 2. Press "Down" to browse different menu options. Press "OK" to select an entry.
- 3. In the menu option, select "Back" to go back to the previous menu.
- 4. The LCD will return to default display after being idle in the menu for longer than 20 seconds.

The following table shows the LCD menu options.

View	Critical Events
Events	Other Events
	Hardware: Hardware version number
	Software: Software version number
Device Info	• P/N : Part number
	MAC: Device MAC address
	Uptime: System up time since the last reboot
Networ	LAN Mode: DHCP, Static IP or PPPoE
k Info	LAN IP: IP address
	LAN Subnet Mask
Networ	LAN Mode: Select LAN mode as DHCP, Static IP or PPPoE
k Menu	Static Routes Reset: Click to reset the static route setting
	○ Reboot
	Factory Reset
	LCD Test Patterns
	Press "OK" to start. Then press the "Down" button to test different LCD patterns. When done, press the "OK" button to exit.
	○ Fan Mode
	Select "Auto" or "On".
	LED Test Patterns
Factory Menu	Select "All On" "All Off" or "Blinking" and check the LED status for USB, SD, T1/E1/J1, Phone 1/Phone 2, Line 1/Line 2 ports. After the test, select "Back" in the menu, and the device will show the LED actual status again.
	RTC Test Patterns
	Select "2022-02-22 22:22" or "2011-01-11 11:11" to start the RTC (Real-Time Clock) test pattern. Check the system time from LCD idle screen by pressing the "DOWN" button, or from the Web GUI->System Status->General page. After the test, reboot the device man and the device will display the correct time.
	Hardware Testing
	Select "Test DSP" to perform the DSP test on the device. This is mainly for factory testing purposes
	which verifies the hardware connection inside the device. The diagnostic result will display on the LCD after the test is done.
Default Passwor d	Showing the default Web login password. Once the password was changed, this menu will not show again.
Web	Protocol: Web access protocol. HTTP or HTTPS. By default, it's HTTPS
Info	Port: Web access port number. By default, it's 8089
	Enable SSH: Enable SSH access.
SSH Switch	Disable SSH: Disable SSH access.
	By default, SSH access is disabled.

Table 4: LCD Menu Options

Use the LED Indicators

The GXW450X has LED indicators in the front to display the connection status. The following table shows the status definitions.

LED Indicator	LED Status
Power	Solid: Connected OFF: Disconnected
WAN	• OFF : Disconnected
T1/E1/J1	 Solid: Connected and working Blinking: No cable is connected; or connected but the link is not working at all.

Table 5: GXW450X LED Indicators

Configuring GXW450X via Web GUI

Web GUI Access

The GXW450X embedded Web server responds to HTTP/HTTPS GET/POST requests. Embedded HTML pages allow users to configure the device t a Web browser such as Microsoft IE, Mozilla Firefox, or Google Chrome.

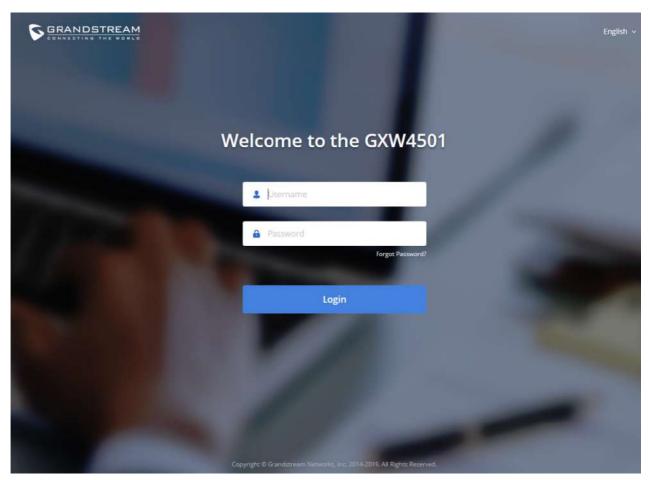


Figure 3: GXW450X Web GUI Login Page

To access the Web GUI:

- 1. Connect the computer to the same network as the GXW450X.
- 2. Ensure the GXW450X is properly powered on and displays the IP address on the LCD screen.
- 3. Open a web browser on the computer and enter the displayed IP address into the search bar in the following format: https://ipaddress:portnumber
- 4. Enter username and password to login. (The default administrator username is "admin" and the default random password can be found on th sticker on the GXW450X).

Reset Password at First Login

At first login, users will be forced to change the default admin password. The following screen will be shown, enter the requested information to p

- o Enter New Password
- o Re-enter New Password
- o **Email Addresses:** Email to be used to recover the password if lost.

Press "Confirmed" to apply the settings and access the web GUI.

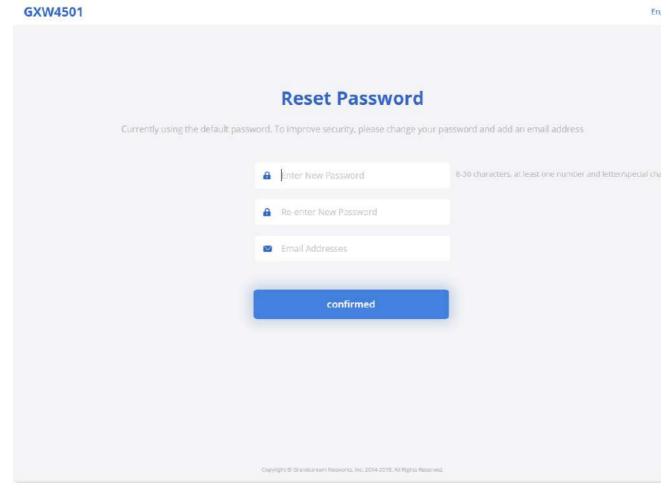


Figure 4: Reset Password

Setup Wizard

When the user logs in to the GXW450X Web GUI for the first time, he will be asked to change the default password and add an email address to ir security, and a setup wizard will provide guidance to set up basic configuration. Configurations in the setup wizard include Network settings, Time and Trunk/routes.

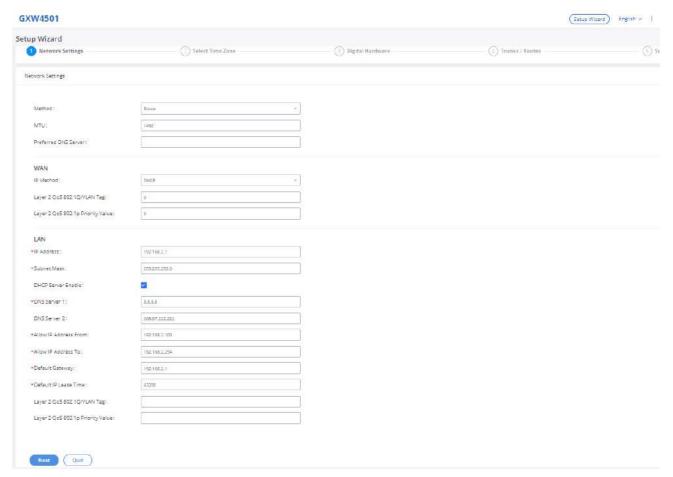


Figure 5: GXW450X Setup Wizard

Web GUI Configurations

There are six main sections in the Web GUI for users to view the Gateway status and configure and manage the GXW450X.

- o System Status: Displays GXW450X Dashboard, System Information, Active calls, and network status.
- o Trunk: To Digital and VoIP trunks and manage inbound/outbound call routes.
- o Gateway Settings: SIP Settings, RTP Settings, and interface settings.
- o System Settings: To configure The HTTP server, network settings, OpenVPN®, security settings, Email Settings, and Time Settings.
- Maintenance: To perform the firmware upgrade, backup configurations, user management cleaner setup, reset/reboot, Syslog setup, and troubleshooting
- o CDR: View call records and download CDR reports.

Web GUI Languages

Currently the GXW450X series Web GUI supports English, Simplified Chinese, Traditional Chinese,

Spanish, French, Portuguese, Russian, Italian, Polish, German, etc.

Users can select the displayed language on the Web GUI login page or at the upper right tab of the Web GUI after logging in.





Figure 6: GXW450X Web GUI Languages

Save and Apply Changes

Click on the "Save" button after configuring the Web GUI options on one page. After saving all the changes, make sure to click on the "Apply Char button on the upper right of the web page to submit all the changes. If the change requires a reboot to take effect, a prompted message will pop you to reboot the device.

SYSTEM STATUS

The System Status section is the interface that allows users to check the general information about the GXW450X such as software and hardware information, space usage, resources usage, etc.

Dashboard

The GXW450X monitors the status of Trunks, Digital Channels, Disk capacities, etc. It presents administrators with the real-time status in different sunder the Web GUI->System Status->Dashboard.

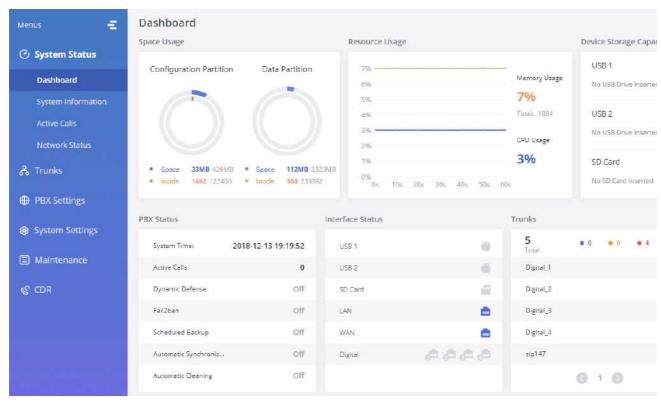


Figure 7: GXW450X Dashboard

Space Usage

Users could access the space usage information from Web GUI→System Status→Dashboard →Space Usage. It shows the available and used space Usage and Inode Usage.

Space Usage includes:

- Configuration partition: This partition contains GXW450X system configuration files and service configuration files.
- o Data partition: CDR records, Voice Prompts, etc.

Inode Usage includes:

- o Configuration partition
- o Data partition

Note: Inode is the pointer used for file reference in the system. The system usually has limited resources of pointers.

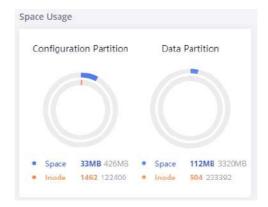


Figure 7: Space Usage

Resource Usage

When configuring and managing the GXW450X, users could access resource usage information to estimate the current usage and allocate the res accordingly. Under Web GUI-System Status-Dashboard -Resource Usage, the current CPU usage and Memory usage are shown in this chart

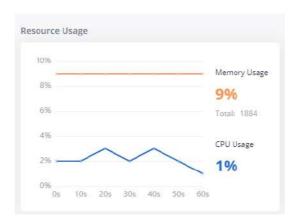


Figure 8: Resource Usage

Disk Capacity

Users could check the external devices' capacities from the Dashboard page of the GXW450X under Web GUI→System Status→Dashboard →De Storage Capacity.



Figure 9: Device Storage Capacity

PBX Status

The PBX status shows the status of some of the gateway GXW450X services. Among the services monitored on the PBX status tab, there is System Active Calls, Schedule backup, etc.



Figure 10: PBX Status

Interfaces Status

This section displays the interface connection status on the GXW450X for USB, SD Card, LAN, WAN, and Digital interfaces.



Figure 11: Interface Status

Trunks

Users could see all the configured trunks' status in this section.



Figure 12: Trunks Status

Four statuses are possible for any trunk configured on the GXW450X:

- o Available
- o Busy
- o Abnormal
- Unmonitored

To visualize the state of each channel of the Digital trunk, users can waver the mouse over the status of the digital trunk as shown on the figure be

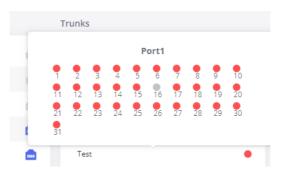


Figure 13: Digital Trunk Channels Status

System Information

The GXW450X system Information can be accessed via Web GUI->System Status->System Information, which displays the following system information.

General

On this menu, users could check the hardware and software information for the GXW450X. Please see the details in the following table.

System Information	System Information			
Model	Product model.			
Part Number	Product part number.			
System Time	Current system time. The current system time is also available on the upper right of each web page.			
Up Time	System up time since the last reboot.			
Version Information				
Boot	Boot version.			
Core	Core version.			
Base	Base version.			

Program	Program version. This is the main software release version.		
Recovery	Recovery version.		

Table 6: System Information → General

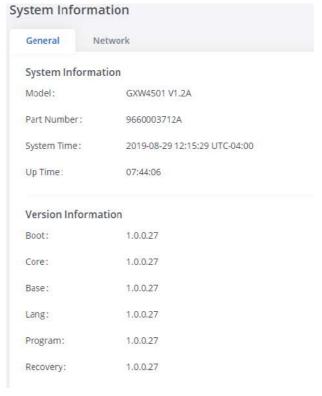


Figure 14: System Information → General

Network

Under Web GUI -> System Status -> System Information -> Network, users could check the network information for the GXW450X. Please see the cin the following table.

WAN/LAN	
MAC Address	Global unique ID of the device, in HEX format. The MAC address can be found on the label coming with the original box and or label located at the bottom of the device.
IPv4 Address	The IPv4 address attributed to the network interface
IPv6 Address	The IPv6 address attributed to the network interface
IPv6 Address Link	The IPv6 address Link attributed to the network interface
Gateway	Default gateway address.
Subnet Mask	Subnet mask address.
DNS Server	DNS server address.

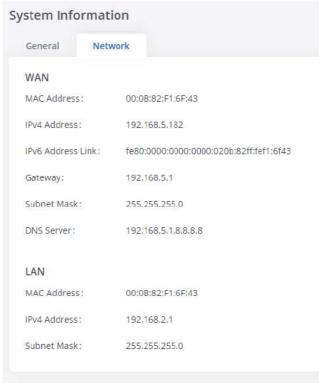


Figure 15: System Information → Network

Active Calls

The active calls on the GXW450X are displayed on the Web GUI **System Status Active Calls** page. Users can monitor call status and hang up a call(s) in a real-time manner.



Figure 16: Active Calls

Users can click on "Hang up All" to terminate all the active calls at once.

Network status

GXW450X supports Network Status to display active internet connections. Users can use Network Status to troubleshoot connection issues betwee GXW450X and other services. This information can be found under Web GUI->System Status->Network Status, the users can view active Interne connections and the Active Unix Domain Sockets.

Network Status					
Active Connections	Active Unix Doma	in Sockets			
Proto	Recv-Q	Send-Q	Local-Address	Foreign-Address	State
tcp	0	0	0.0.0.0:8088	0.0.0.0:*	LISTEN
tcp	0	0	0.0.0.0:8888	0.0.0.0:*	LISTEN
tcp	0	0	127.0.0.1:25	0.0.0.0:*	LISTEN
tcp	0	0	0.0.0.0:7777	0.0.0.0:*	LISTEN
tcp	0	0	0.0.0.0:7681	0.0.0.0:*	LISTEN
tcp	0	0	127.0.0.1:7777	127.0.0.1:36727	TIME_WAIT
tcp	0	0	127.0.0.1:7777	127.0.0.1;36728	TIME_WAIT
tcp	0	0	127.0.0.1:7681	127.0.0.1:40182	ESTABLISHED
tcp	0	0	127.0.0.1;40182	127.0.0.1:7681	ESTABLISHED

Figure 17: Active connections

Network Status					
Active Connections	Active Unix Doma	in Sockets			
Proto	RefCnt	Flags	Type	State	I-Node
unix	2	[ACC]	SEQPACKET	LISTENING	9226
unix	9	О	DGRAM		11548
unix	2	[ACC]	STREAM	LISTENING	1922
unix	2	[ACC]	STREAM	LISTENING	10371
unix	2	0	DGRAM		10384
unix	2	[ACC]	STREAM	LISTENING	12486
unix	2	[ACC]	STREAM	LISTENING	13150

Figure 18: Active Unix Domain Sockets

SYSTEM SETTINGS

This chapter explains configurations for system-wide parameters on the GXW450X. System settings are under the "System Settings" tab on GXW4! Web GUI. System settings include Network Settings, Security Settings, HTTP Server, Email Settings, Time Settings, OpenVPN® settings, and DDNS Settings

HTTP Server

The GXW450X embedded web server responds to HTTP/HTTPS GET/POST requests. Embedded HTML pages allow the users to configure the gate through a Web browser such as Microsoft IE, Mozilla Firefox, and Google Chrome. By default, the Gateway can be accessed via HTTPS using Port & (e.g., https://192.168.1.50:8089). Users could also change the access protocol and port as preferred under Web GUI→System Settings→HTTP Set

Basic Settings	
Redirect From Port 80	Enable or disable redirect from port 80. On the gateway, the default access protocol is HTTPS and the default port number is 8089. When this option is enabled, the access using HTTP with Port 80 will be redirected to HTTPS with Port 8089. The default setting is "Enable".
Protocol Type	Select HTTP or HTTPS. The default setting is "HTTPS". This is also the protocol used for zero config when the endpoint device downloads the config file from the GXW450X.
Port	Specify the port number to access the HTTP server. The default port is 8089.

Enable IP whitelist	If enabled, only the IP address on the permitted IP list will be allowed to access the GXW450X's web GUI.
Permitted IP(s)	Add an IP address to the list of allowed IPs to access GXW450X's web GUI. Ex: 192.168.6.233 / 255.255.255.255
Certificate Settings	
Options	Select the mode to download SSL certificates for the web server, two modes are available: • Manually Upload certificate: Upload the files while respecting size and format. • Automatically request certificate: enter the domain from which to request the certificate files.
TLS Private Key	Upload private key for the built-in HTTP server. Note: The size of the key file must be under 2MB and it will be renamed as "private.pem" automatically.
TLS Cert	Upload the certificate for the built-in HTTP server and override the existing one. Note: The size of your certificate must be under 2MB. This is the certificate file (*.pem format only) for TLS connection and it will be renamed as "certificate.pem" automatically. It contains a private key for the client and a signed certificate for the server.
Reset Certificate	Restore the default key and certificate. The web server needs to reload to take effect after certificate restoration.

Network Settings

After successfully connecting the GXW450X to the network for the first time, users could log in to the Web GUI and go to System Settings Network Settings to configure the network parameters for the device. In this section, all the available network setting options are listed. Select each tab on Web GUI >System Settings Network Settings page to configure IPV4 Settings, IPV6 Settings, 802.1X and Static Routes.

Basic Settings

Please refer to the following tables for basic network configuration parameters on GXW450X.

Method	Switch : WAN port interface will be used for the uplink connection. LAN port interface be used as a room for PC connection.
МТИ	Specifies the Maximum Transmission Unit. (By default, it's 1492)
IPv4 Address	
Preferred DNS Server	Enter the preferred DNS server address. If Preferred DNS is used, GXW450X will try to as the Primary DNS server.
WAN (when Method set to "Route") / LAN (when Method set to "Switch")	
IP Method	Select DHCP, Static IP, or PPPoE. The default setting is DHCP.
If "IP Method" is set to "Static"	
IP Address	Enter the IP address for static IP settings. The default setting is 192.168.0.160
Subnet Mask	Enter the subnet mask address for static IP settings. The default setting is 255.255.0.0.
Gateway IP	Enter the gateway IP address for static IP settings. The default setting is 0.0.0.0

DNS Server 1	Enter the DNS server 1 address for static IP settings.			
DNS Server 2	Enter the DNS server 2 address for static IP settings.			
If "IP Method" is set to "PPPoE"				
User Name	Enter the user name to connect via PPPoE.			
Password	Enter the password to connect via PPPoE.			
If "IP Method" is set to "DHCP", "Static" or "PPP	PoE"			
Layer 2 QoS 802.1Q/VLAN Tag	Assign the VLAN tag of the layer 2 QoS packets for the LAN port. The default value is VLAN Tag). The valid range is between 2 and 4094.			
Layer 2 QoS 802.1p Priority Value	Assign the priority value of the layer 2 QoS packets for the LAN port. The default value The valid range is between 0 and 7.			
LAN (when Method set to "Route")				
IP Address	Enter the IP address. The default setting is 192.168.2.1			
Subnet Mask	Enter the subnet mask address. The default setting is 255.255.255.0.			
DHCP Server Enable	Enable or disable DHCP server capability. The default setting is "Yes.			
DNS Server 1	Enter DNS server address 1. The default setting is 8.8.8.8			
DNS Server 2	Enter DNS server address 2. The default setting is 208.67.222.222.			
Allow IP Address From	Enter the DHCP IP Pool starting address. The default setting is 192.168.2.100.			
Allow IP Address To	Enter the DHCP IP Pool ending address. The default setting is 192.168.2.254.			
Default Gateway	Configure the default Gateway assigned by the DHCP server.			
Default IP Lease Time	Enter the IP lease time (in seconds). The default setting is 43200.			
Layer 2 QoS 802.1Q/VLAN Tag	Assign the VLAN tag of the layer 2 QoS packets for the LAN port. The default value is VLAN Tag). The valid range is between 2 and 4094.			
Layer 2 QoS 802.1p Priority Value	Assign the priority value of the layer 2 QoS packets for LAN port. The default value is valid range is between 0 and 7.			
IPv6 Address				
WAN (when "Method" is set to "Route") / LAN (when "Method" is set to "Switch")				
IP Method	Select Auto or Static. The default setting is Auto			
If "IP Method" is set to "Static"				
IP Address	Enter the IP address for static IP settings.			
IP Prefixlen	Enter the Prefix length. Default is 64			
DNS Server 1	Enter the DNS server 1 address for static settings.			
DNS Server 2	Enter the DNS server 2 address for static settings.			

LAN (when "Method" is set to "Route")		
DHCP Server	Enable or disable DHCP server capability. Available options are: • Disable: DHCP Server will be disabled • Auto: Stateless address auto-configuration using NDP protocol • DHCPv6: Stateful address autoconfiguration using DHCPv6 protocol. The default setting is "Disable"	
If "DHCP Server" is set to "Auto" or "DHCPv6"		
DHCP Prefix	Enter DHCP Prefix when static IP is used. Format: "xxxx:xxxx:xxxx:xxxx:xxxx:xxxx:xxxx:x	
DHCP Prefixlen	Enter the Prefix length. Default is 64	
DNS Server 1	Enter DNS server address 1. The default setting is "2001:4860:4860:8888".	
DNS Server 2	Enter DNS server address 2. The default setting is "2001:4860:4860::8844".	
If "DHCP Server" is set to "DHCPv6"		
Allow IP Address From	Enter the DHCP IP Pool starting address. The default setting is "2001:db8:2:2::3000".	
Allow IP Address To	Enter the DHCP IP Pool ending address. The default setting is "2001:db8:2:2::4000".	
Default IP Lease Time	Enter the IP lease time (in seconds). The default setting is 43200.	

Table 7: GXW450X Network Settings →Basic Settings

DHCP Client List

This feature can bind MAC to IP address on the LAN port when GXW450x is set to "Route" mode.

When devices receive IP addresses from the GXW450X LAN port, they will be listed on the web UI under "System Settings → Network Settings · DHCP Client List" as shown below.



Figure 19: DHCP Client List

Users can bind manually a MAC to an IP address by clicking on + Add Bind New MAC Address, the following figure will pop up.

Figure 20: Add Bind New MAC Address Bind

The user needs to set the device MAC address and the IP that will be bound to it (the IP address needs to be within the GXW450X DHCP range).

In order to bind a batch of listed MAC addresses, the user needs to check first the MAC addresses to bind and click on popup will be shown, click ox to bind the addresses.

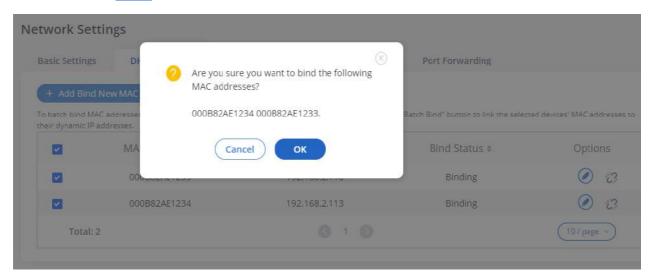


Figure 21: Batch Add MAC Address Bind

After Clicking "OK" to confirm the binding, the "Bind Status" will change from "Unbind" to "Binding".

802.1X Settings

IEEE 802.1X is an IEEE standard for port-based network access control. It provides an authentication mechanism to the device before the device ca access the Internet or other LAN resources. The GXW450X supports 802.1X as a supplicant/client to be authenticated. The following diagram and show the GXW450X uses 802.1X mode "EAP-MD5" on the WAN port as the client in the network to access the Internet.

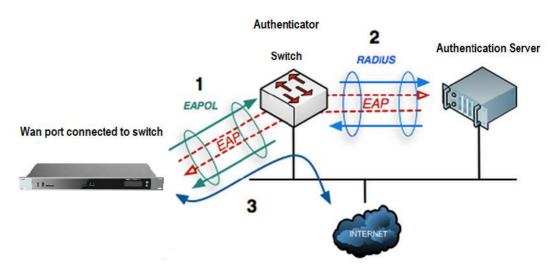


Figure 22: GXW450X Using 802.1X as Client

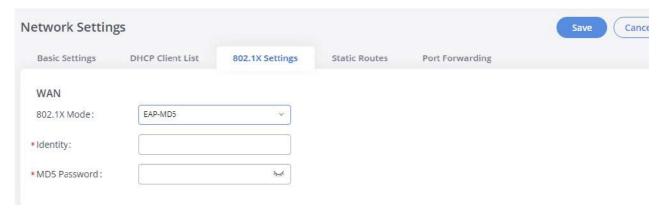


Figure 23: GXW450X using 802.1X EAP-MD5

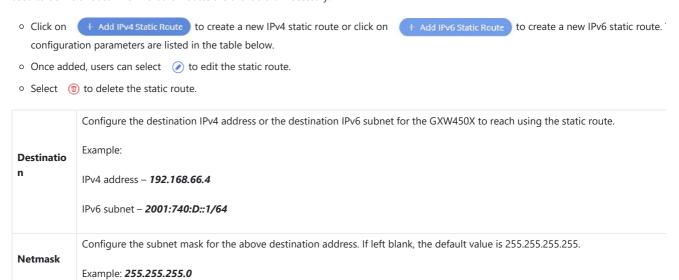
The following table shows the configuration parameters for 802.1X on GXW450X. Identity and MD5 password are required for authentication, which should be provided by the network administrator obtained from the RADIUS server. If "EAP-TLS" or "EAP-PEAPv0/MSCHAPv2" is used as the 802.1 mode, users will also need to upload 802.1X CA Certificate and 802.1X Client Certificate, which should be also generated from the RADIUS server.

802.1X Mode	Select 802.1X mode. The default setting is "Disable". The supported 802.1X mode is: • EAP-MD5 • EAP-TLS • EAP-PEAPv0/MSCHAPv2
Identity	Enter 802.1X mode Identity information.
MD5 Password	Enter 802.1X mode MD5 password information.
802.1X CA Certificate	Upload 802.1X CA certificate. This file will be renamed as "8021x_ca_cert" automatically.
802.1X Client Certificate	Upload 802.1X client certificate with both certificate and private key. This file will be renamed as "8021x_client_cert" automatically.

Table 8: GXW450X Network Settings →802.1X

Static Routes

The GXW450X provides users static routing capability that allows the device to use manually configured routes, rather than information only from dynamic routing or gateway configured in the GXW450X Web GUI->System Settings->Network Settings->Basic Settings to forward traffic. It ca used to define a route when no other routes are available or necessary.



	Configure the IPv4 or IPv6 gateway address so that the GXW450X can reach the destination via this gateway. The gateway address optional.
Gateway	Example:
	192.168.40.5 or 2001:740:D::1
Interface	Specify the network interface on the GXW450X to reach the destination using the static route.

Table 9: GXW450X Network Settings → Static Routes

Port Forwarding

The GXW450X network interface supports router functions which provide users the ability to do port forwarding. If the GXW450X is set to "Route" Web GUI->System Settings->Network Settings->Basic Settings: Method, port forwarding is available for configuration.

The port forwarding configuration is under the Web GUI→System Settings→Network Settings→Port Forwarding page. Please see the related setting the table below.

	Specify the WAN port number or a range of WAN ports. An unlimited number of ports can be configured.
WAN Port	Note: When it is set to a range, the WAN port, and LAN port must be configured with the same range, such as WAN port: 1000-1005 and LA port: 1000-1005, and access from the WAN port will be forwarded to the LAN port with the same port number, for example, WAN port will be port forwarding to LAN port 1000.
LAN IP	Specify the LAN IP address.
LAN Port	Note: When it is set to a range, the WAN port, and LAN port must be configured with the same range, such as WAN port: 1000-1005 and LA port: 1000-1005, and access from the WAN port will be forwarded to the LAN port with the same port number, for example, WAN port will be port forwarding to LAN port 1000.
Protoc ol Type	Select protocol type "UDP Only", "TCP Only" or "TCP/UDP" for the forwarding in the selected port. The default setting is "UDP Only".

Table 10: GXW450X System Settings → Network Settings →Port Forwarding

OpenVPN®

OpenVPN® settings allow the users to configure GXW450X to use VPN features, the following table gives details about the various options in ord configure the GXW450X as OpenVPN Client.

OpenVPN® Enable	Enable / Disable the OpenVPN® feature. The default is "Disabled".	
Configuration Method	Select the OpenVPN® configuration method. • Manual Configuration. • Upload Configuration File.	
If "Configuration Method" is set to "Manual Configuration"		
OpenVPN® Server Address	Configures the hostname/IP and port of the server. For example, "192.168.1.2:22" or "2001:0DB8:0000:0000:0000:1428:0000".	

OpenVPN® Server Protocol	Select the same protocol that the OpenVPN® server is using, e.g., select UDP if the OpenVPN® i using UDP. Available options: • UDP • TCP The default setting is "UDP".
OpenVPN® Device Mode	Use the same setting as used on the server. • Dev TUN: Create a routed IP tunnel. • Dev TAP: Create an Ethernet tunnel. The default setting is "Dev TUN".
OpenVPN® Use Compression	Compress tunnel packets using the LZO algorithm on the VPN link. Don't enable this unless it is a enabled in the server config file.
OpenVPN® Encryption Algorithm	Please select a cryptographic cipher from the drop-down list. Use the same setting that you are to on the server. The default setting is "BF-CBC(Blowfish)".
OpenVPN® CA Cert	Upload an SSL/TLS root certificate. This file will be renamed as 'ca.crt' automatically.
OpenVPN® Client Cert	Upload a client certificate. This file will be renamed as 'client.crt' automatically.
OpenVPN® Client Key	Upload a client private key. This file will be renamed as 'client.key' automatically.
User Authentication	Enables the authentification by entering the Username and Password credentials , Disabled by Default.
If "Configuration Method" is set to "Manual Configuration"	
OpenVPN® Configuration File	Upload Configuration file to with OpenVPN® settings. Only file with .conf,.ovpn suffix is accepted for OpenVPN® Configuration File. The file size must under 2MB.

Table 11: GXW450X System Settings →Network Settings →OpenVPN®

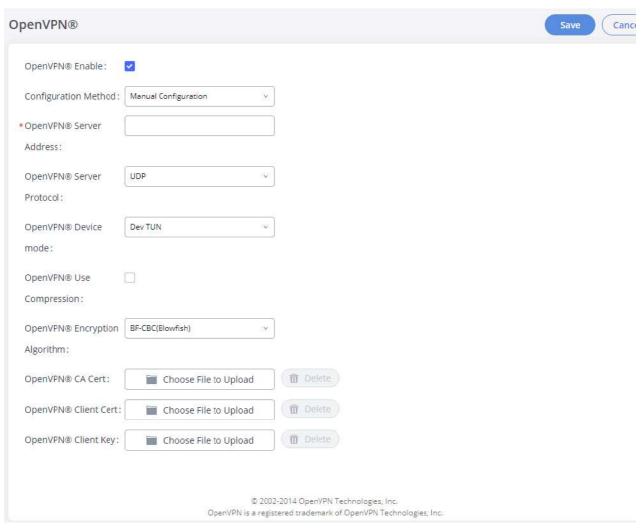


Figure 24: OpenVPN® Feature on the GXW450X

DDNS Settings

DDNS setting allows users to access GXW450X via domain name instead of IP address.

The GXW450X supports DDNS service from the following DDNS provider:

- o dydns.org
- o freedns.afraid.org
- o zoneedit.com
- o noip.com
- o oray.net

Here is an example of using noip.com for DDNS.

1. Register domain in DDNS service provider. Please note the GXW450X needs to have public IP access.

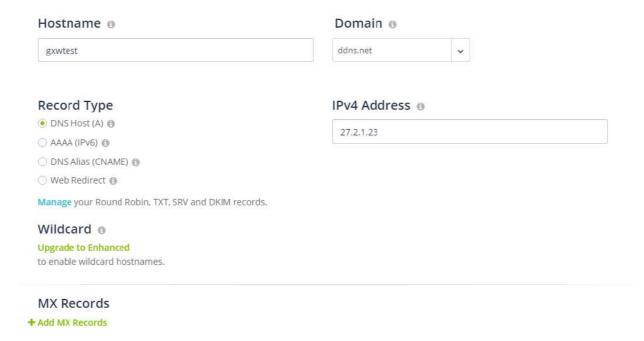


Figure 25: Register Domain Name on Noip.com

2. On Web GUI -> System Settings -> Network Settings -> DDNS Settings, enable DDNS service and configure username, password, and hostnar

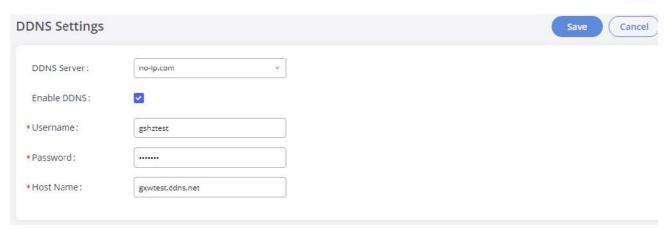


Figure 26: GXW450X DDNS Settings

3. Now you can use a domain name instead of an IP address to connect to the GXW450X Web GUI.

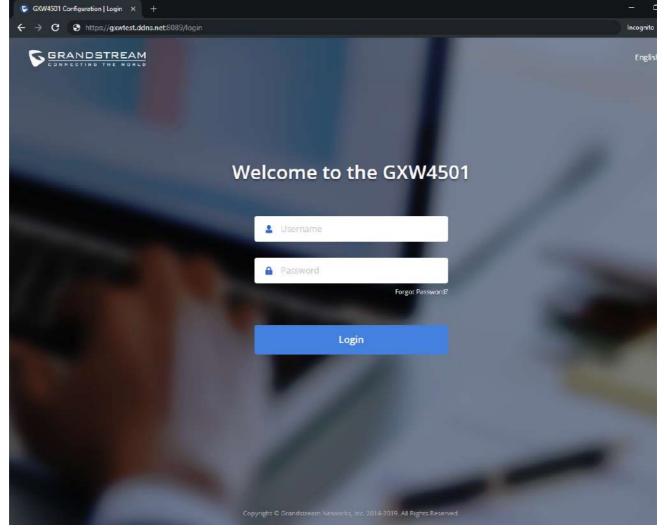


Figure 27: Using Domaine Name to Connect to GXW450X

Security Settings

The GXW450X provides users with firewall security configurations to prevent certain malicious attacks on the GXW450X system. Users could configuration to prevent certain malicious attacks on the GXW450X system. Users could configuration, restrict or reject specific traffic through the device for security and bandwidth purpose. The GXW450X also provides the Fail2ban feature for authentication errors in SIP REGISTER, INVITE and SUBSCRIBE. To configure firewall settings in the GXW450X, go to the Web GUI→System Settings→Security Settings page.

Static Defense

Under the Web GUI->System Settings->Security Settings->Static Defense page, users will see the following information:

- o Current service information with port, process, and type.
- o Custom firewall settings.
- o Typical firewall settings.

The following table shows a sample current service status running on the GXW450X.

Port	Process	Туре	Port	Process	Туре
2000	asterisk	TCP/IPv4	67	dhcpd	UDP/IPv4
8088	asterisk	TCP/IPv4	69	udpsvd	UDP/IPv4
8888	pbxmid	TCP/IPv4	37178	asterisk	UDP/IPv4
25	master	TCP/IPv4	80	lighttpd	TCP/IPv6
7777	asterisk	TCP/IPv4		master	TCP/IPv6

Port	Process	Туре	Port	Process	Туре
7681	pbxmid	TCP/IPv4	8089	lighttpd	TCP/IPv6
4520	asterisk	UDP/IPv4	4569	asterisk	UDP/IPv6
4569	asterisk	UDP/IPv4	5060	asterisk	UDP/IPv6
3765	dhcpd	UDP/IPv4	24539	dhcpd	UDP/IPv6
5000	asterisk	UDP/IPv4	54411	asterisk	UDP/IPv6
67	udhcpd	UDP/IPv4			

Table 12: GXW450X Static Defense → Current Service

Under "Custom Firewall Settings", users could create new rules to accept, reject or drop certain traffic going through the GXW450X. To create a ne click on the "Create New Rule" button and a new window will pop up for users to specify rule options.

Right next to the "Create New Rule" button, there is a checkbox for the option "Reject Rules". If it's checked, all the rules will be rejected except th firewall rules listed below. In the firewall rules, only when there is a rule that meets all the following requirements, the option "Reject Rules" will be allowed to check:

- o Action: "Accept"
- ∘ Type "In"
- o The destination port is set to the system login port (e.g., by default 8089)
- The protocol is not UDP

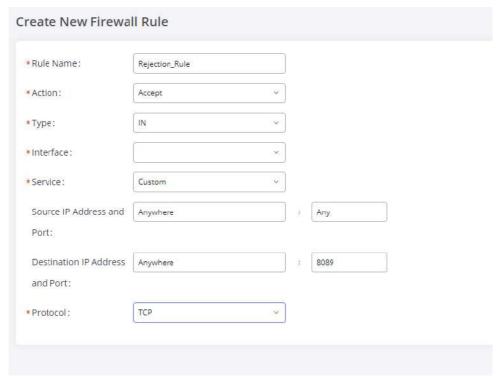


Figure 28: Create New Firewall Rule

Below is a table listing all the firewall rules settings:

Rule Name	Specify the Firewall rule name to identify the firewall rule.
Action	Select the action for the Firewall to perform. 1. ACCEPT 2. REJECT 3. DROP
Туре	Select the traffic type.

	 IN: If selected, users will need to specify to the network interface (for GXW450X) for the incoming traffic, the network interface can be set to "WAN", "LAN", or Both. OUT
Service	Select the service type. 1. FTP 2. SSH 3. Telnet 4. HTTP 5. Custom
	If "Custom" is selected, users will need to specify Source (IP and port), Destir (IP and port), and Protocol (TCP, UDP, or Both) for the service. Please note if source or the destination field is left blank, it will be used as "Anywhere".

Table 13: Firewall Rule Settings

Save the change and click on the "Apply" button. Then submit the configuration by clicking on "Apply Changes" on the upper right of the web page new rule will be listed at the bottom of the page with sequence number, rule name, action, protocol, type, source, destination, and operation. Mor operations are below:

- Click on o to edit the rule.
- o Click on to delete the rule.
- o Use the arrows up ⊗ ,down ⊚ , to the top ⊚ or to the bottom ⊗ to move the rules up and down.

For typical firewall settings, users could configure the following options on the GXW450X.

Ping Defense Enable	If enabled, ICMP response will not be allowed for Ping requests. The default setting is disabled. To enable or disable it, clic the check box for the LAN or WAN (GXW450X) interface.
SYN-Flood Defense Enable	Allows the GXW450X to handle excessive amounts of SYN packets from one source and keep the web portal access. There two options available and only one of these options may be enabled at one time. o eth(0)LAN defends against attacks directed to the LAN IP address of the GXW450X. o eth(1)WAN defends against attacks directed to the WAN IP address of the GXW450X. SYN Flood Defense will limit the number of SYN packets accepted by the GXW450X from one source to 10 packets per sec Any excess packets from that source will be discarded.
Ping-of-Death Defense Enable	Enable to prevent Ping-of-Death attack on the device. The default setting is disabled. To enable or disable it, click on the c box for the LAN or WAN (GXW450X) interface.

Table 14: Typical Firewall Settings

Dynamic Defense

Dynamic defense is supported on the GXW450X series. It can blacklist hosts dynamically when the LAN mode is set to "Route" under the Web GUI→System Settings→Basic Settings page. If enabled, the traffic coming into the GXW450X can be monitored, which helps pr massive connection attempts or brute force attacks on the device. The blacklist can be created and updated by the GXW450X firewall, which will tl displayed on the web page. Please refer to the following table for dynamic defense options on the GXW450X.

Dynamic Defense Enable	Enable dynamic defense. The default setting is disabled.
Blacklist Update Interval	Configure the blacklist update time interval (in seconds). The default setting is 120.
Connection Threshold	Configure the connection threshold. Once the number of connections from the same host reaches the threshold, it will b added to the blacklist. The default setting is 100.

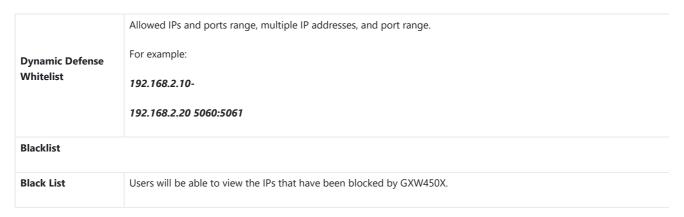


Table 15: GXW450X Firewall Dynamic Defense

The following figure shows a configuration example:

- If a host at IP address 192.168.2.5 initiates more than 100 TCP connections to the GXW450X, it will be added to the GXW450X blacklist. This had 192.168.2.5 will be blocked by the GXW450X for 500 seconds.
- Since IP range 192.168.2.10-192.168.2.20 is in the whitelist, if a host initiates more than 20 TCP connections to the GXW450X within 1 minute, not be added to the GXW450X blacklist. It can still establish a TCP connection with the GXW450X.

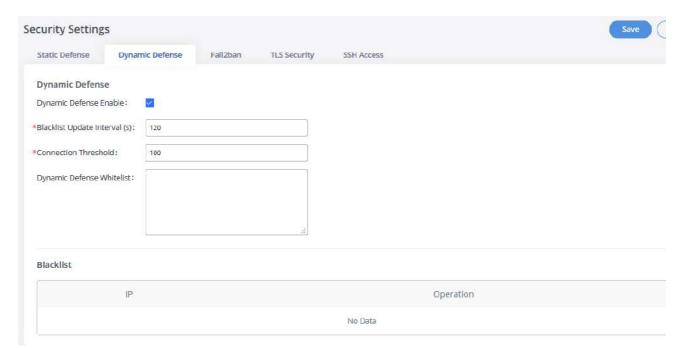


Figure 29: Dynamic Defense Configuration

Fail2Ban

Fail2Ban feature on the GXW450X provides intrusion detection and prevention for authentication errors in SIP INVITE and SUBSCRIBE. Once the er detected within "Max Retry Duration", the GXW450X will act to forbid the host for a certain period as defined in "Banned Duration". This feature he prevent SIP brute force attacks on the gateway system.

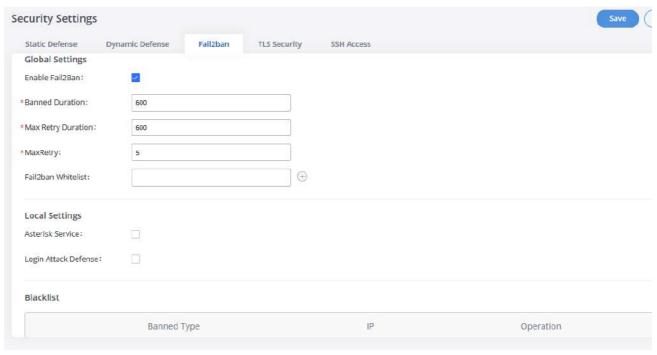


Figure 30: Fail2Ban Settings

Global Settings	
Enable Fail2Ban	Enable Fail2Ban. The default setting is disabled. Please make sure both "Enable Fail2Ban" and "Asterisk Service" are turned on to Fail2Ban for SIP authentication on the GXW450X.
Banned Duration	Configure the duration (in seconds) for the detected host to be banned. The default setting is 600. If set to 0, the host will be alv banned.
Max Retry Duration	Within this duration (in seconds), if a host exceeds the max times of retry as defined in "MaxRetry", the host will be banned. The default setting is 600.
MaxRetry	Configure the number of authentication failures during "Max Retry Duration" before the host is banned. The default setting is 5.
Fail2Ban Whitelist	Configure IP address, CIDR mask, or DNS host in the whitelist. Fail2Ban will not ban the host with a matching address in this list. 20 addresses can be added to the list.
Local Settings	
Asterisk Service	Enable Asterisk service for Fail2Ban. The default setting is disabled. Please make sure both "Enable Fail2Ban" and "Asterisk Servic turned on to use Fail2Ban for SIP authentication on the GXW450X.
Listening Port Number	Configure the listening port number for the service. By default, port 5060 will be used for UDP and TCP, and port 5061 will be used for UDP and TCP, and port 5061 will be used for UDP and TCP.
MaxRetry	Configure the number of authentication failures during "Max Retry Duration" before the host is banned. The default setting is 10 Please make sure this option is properly configured as it will override the "MaxRetry" value under "Global Settings".
Login Attack Defense	Enables defense against excessive login attacks to the GXW450X's web GUI. The default setting is disabled.
Listening Port Number	This is the Web GUI listening port number which is configured under System Settings >HTTP Server >Port . The default is 808!
MaxRetry	When the number of failed login attempts from an IP address exceeds the MaxRetry number, that IP address will be banned from accessing the Web GUI. The default setting is 5
Blacklist	

Table 16: Fail2Ban Settings

TLS Security

Under the Web **GUI→System Settings→Security Settings→TLS security** page, users can now select the minimum and maximum versions of TLS GXW450x to support.

Maximum TLS Version	Specifies the minimum TLS version on the GXW450x in order to accept TLS connection.
Minimum TLS Version	Specifies the maximum TLS version on the GXW450x in order to accept TLS connection.

Table 17: TLS Security parameters



Figure 31: TLS Security

SSH Access

SSH switch is available via Web GUI. Users can enable or disable SSH access directly from the Web GUI or LCD screen. For web SSH access, please GXW450X web interface and go to Web GUI ->System Settings->Security Settings->SSH Access. By default, SSH access is disabled for security concerns. It is highly recommended to only enable SSH access for debugging purposes



Figure 32: SSH Access

Time Settings

Automatic Date and Time

The current system time on the GXW450X can be found under Web GUI→System Status→Dashboard→PBX Status.

To configure the GXW450X to update the time automatically, go to Web GUI->System Settings->Time Settings->Automatic Date and Time.

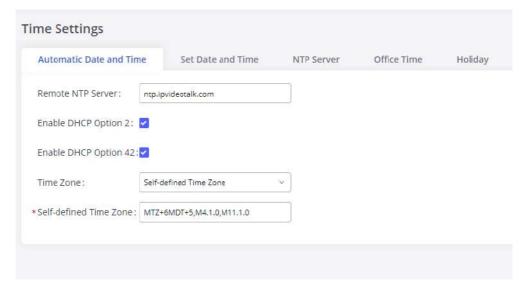


Figure 33: Automatic Date and Time Settings

The configurations under Web GUI→Settings→Time Settings→Automatic Date and Time page require reboot to take effect. Please consider configuring Automatic Date and Time-related changes when setting up the GXW450X for the first time to avoid service interruption after installation and deployment production.

Remote NTP Server	Specify the URL or IP address of the NTP server for the GXW450X to synchronize the date and time. The default NTP server is ntp.ipvideotalk.com.
Enable DHCP Option 2	If set to "Yes", the GXW450X can get provisioned for Time Zone from DHCP Option 2 in the local server automatically. The defau setting is "Yes".
Enable DHCP Option 42	If set to "Yes", the GXW450X can get provisioned for NTP Server from DHCP Option 42 in the local server automatically. This will override the manually configured NTP Server. The default setting is "Yes".
	Select the proper time zone option so the GXW450X can display the correct time accordingly.
Time Zone	If "Self-Defined Tome Zone" is selected, please specify the time zone parameters in the "Self-Defined Time Zone" field as describ the below option.
	If "Self-Defined Time Zone" is selected in the "Time Zone" option, users will need to define their own time zone following the for below.
	The syntax is: std offset dst [offset], start [/time], end [/time]
	Default is set to: MTZ+6MDT+5,M4.1.0,M11.1.0
	MTZ+6MDT+5
	This indicates a time zone with 6 hours offset and 1 hour ahead for DST, which is U.S central time. If it is positive (+), the local time zone is west of the Prime Meridian (A.K.A: International or Greenwich Meridian); If it is negative (-), the local time zone is east.
Self-Defined Time Zone	M4.1.0,M11.1.0
	The 1st number indicates Month: 1,2,3, 12 (for Jan, Feb,, Dec).
	The 2nd number indicates the nth iteration of the weekday: (1st Sunday,
	3rd Tuesday). Normally 1, 2, 3, 4 are used. If 5 is used, it means the last iteration of the weekday.
	The 3rd number indicates weekday: 0,1,2,,6 (for Sun, Mon, Tues,, Sat).
	Therefore, this example is the DST which starts on the First Sunday of
	April to the 1st Sunday of November.

Set Date and Time

To manually set the time on the GXW450X, go to Web GUI->System Settings->Time Settings->Set Date and Time. The format is YYYY-MM-DD HH:MM:SS.



Figure 34: Date and Time Manual Configuration

Current Date and Time	Manually set up the system time. If the system time is automatically set up successfully, the manually configured value v not take effect.
Date Format	Configure the global date format, the default format is yyyy-mm-dd.
Time Format	Chooses the format that will be used to display the Time, 24-hour format or 12-hour format, the default setting is the 24 format

Table 19: Date and Time Manual Settings

Manual setup of time will take effect immediately after saving and applying changes in the Web GUI. If users would like to reboot the GXW450X and keep manual setup time setting, please make sure "Remote NTP Server", "Enable DHCP Option 2" and "Enable DHCP Option 42" options under Web GUI->Settings->Time Settings-> Automatic Date and Time page are unchecked or set to empty. Otherwise, time auto updating settings in this page will ta effect after reboot.

NTP Server

The GXW450X can be used as an NTP server for the NTP clients to synchronize their time. To configure the GXW450X as the NTP server, set "Enab server" to "Yes" under Web GUI->System Settings->Time Settings->NTP Server. On the client side, point the NTP server address to the GXW450X address or hostname to use the GXW450X as the NTP server.



Figure 35: GXW450X NTP Server

Office Time

On the GXW450X, the system administrator can define "office time", which can be used to configure time conditions for the inbound rule schedule configure office time, go to Web GUI->System Settings->Time Settings->Office Time. Click on "Add Office Time" to create an office time.

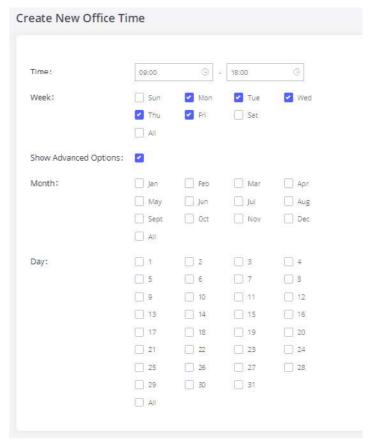


Figure 36: Add New Office Time

Time	Configure the start time and end time for office hours.
Week	Select the work days in one week.
Show Advanced Options	Check this option to show advanced options. Once selected, please specify the "Month" and "Day" options.
Month	Select the months for office time.
Day	Select the work days in one month.

Table 20: Office Time Settings

Select "Time" and the day for the "Week" for the office time. The system administrator can also define the month and day of the month as advanc options. Once done, click on "Save" and then "Apply Change" for the office time to take effect. The office time will be listed on the web page as th figure shows below.

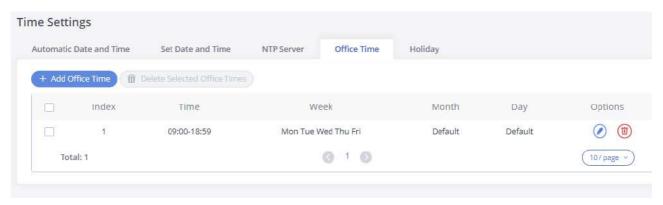


Figure 37: Time Settings →Office Time

- Click on oto edit the office time.
- o Click on to delete the office time.
- o Click on "Delete Selected Office Times" to delete multiple selected office times at once.

On the GXW450X, the system administrator can define "holiday", which can be used to configure time conditions for the inbound rule schedule. To configure holiday, go to Web GUI->System Settings->Time Settings->Holiday. Click on "Add Holiday" to create holiday time.

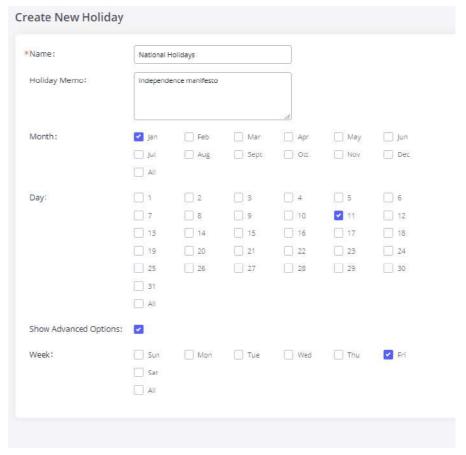


Figure 38: Add a Holiday

Name	Specify the holiday name to identify this holiday.
Holiday Memo	Create a note for the holiday.
Month	Select the month for the holiday.
Day	Select the day for the holiday.
Show Advanced Options	Check this option to show advanced options. If selected, please specify the days as holidays in one week below.
Week	Select the days as holidays in one week.

Table 21: Holiday Settings

Enter holiday "Name" and "Holiday Memo" for the new holiday. Then select "Month" and "Day". The system administrator can also define days in week as advanced options. Once done, click on "Save" and then "Apply Change" for the holiday to take effect. The holiday will be listed in the web as the figure shows below.

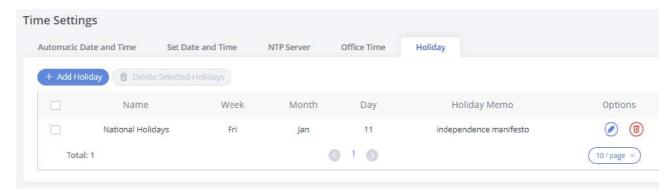


Figure 39: Time Settings → Holiday

- o Click on to delete the holiday.
- o Click on "Delete Selected Holidays" to delete multiple selected holidays at once.

Email Settings

Email Settings

The Email application on the GXW450X can be used to send out alert event Emails, retrieve admin password, etc. The configuration parameters ca accessed via Web GUI->System Settings->Email Settings.

TLS Enable	Enable or disable TLS during transferring/submitting your Email to another SMTP server. The default setting is "Yes".
Туре	 MTA: Mail Transfer Agent. The Email will be sent from the configured domain. When MTA is selected, there is no need to set SMTP server for it, or no user login is required. However, the Emails sent from MTA might be considered spam by the target S server. Client: Submit Emails to the SMTP server. An SMTP server is required, and users need log in with the correct credentials.
Email Template Sending	Select the email template format to be sent. The "HTML" format is compatible with most mail clients and is recommended. If the client does not support the "HTML" format, please select the "Plain Text" format.
Domain	Specify the domain name to be used in the Email when using the type "MTA".
SMTP Server	Specify the SMTP server when using the type "Client".
Enable SASL Authentica tion	Enable SASL Authentication. When disabled, GXW450X will not try to use the user name and password for mail client login authentication. Most of the mail server requires login authentication while some other private mail servers allow anonymous login which requires disabling this option to send Email as normal. For Exchange Server, please disable this option.
Username	A username is required when using the type "Client". Normally it's the Email address.
Password	Password to login for the above Username (Email address) is required when using the type "Client".
POP/POP3 Server Address	Configure the POP/POP3 server address for the configured username Example: pop.gmail.com
POP/POP3 Server Port	Configure the POP/POP3 server port for the configured username Example: 995
Display Name	Specify the display name in the FROM header in the Email.
Sender	Specify the sender's Email address.
Jenuel	For example pbx@example.mycompany.com.

Table 22: Email Settings

The following figure shows a sample Email setting on the GXW450X, assuming the email is using the default SMTP server of Gmail.

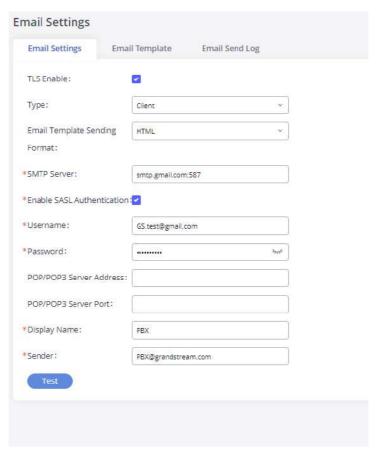


Figure 40: Email Settings

Once the configuration is finished, click on "Test". In the prompt, fill in a valid Email address to send a test Email to verify the Email settings on the GXW450X.

Email Template

The Email templates on the GXW450X can be used for email notification. The configuration parameters can be accessed via Web GUI→System Settings→Email Settings→Email Templates.



Figure 41: Email Templates

- Press C Reset All Template Images to reset all email templates to default ones.

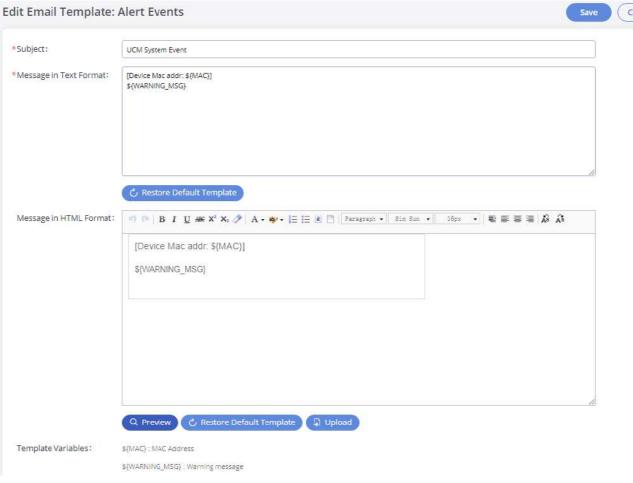


Figure 42: Alert Events Template

- Users can preview mail sample by clicking on Preview .
- Click on Click on Restore Default Template in order to restore the default email template.

Email Send Log

Under GXW450X Web GUI **System Settings Email Settings Email Send Log**, users could search, filter, and check whether the Email is sent o successfully or not. This page will also display the corresponding error message if the Email is not sent out successfully.

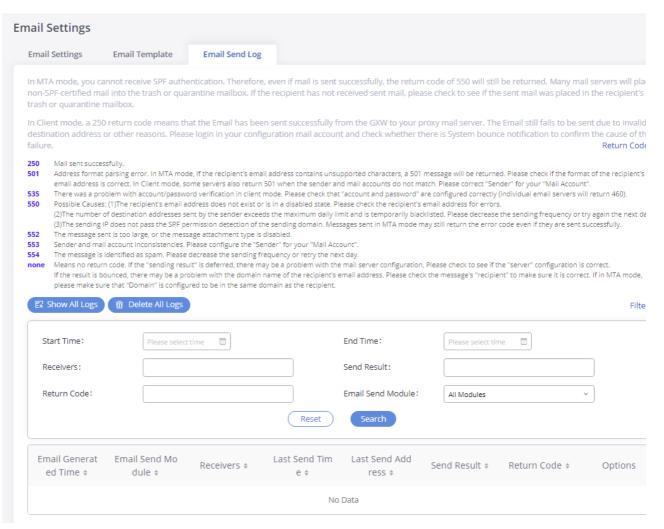


Figure 43: Email Send Log

Field	Description
Start Time	Enter the start time for the filter
End time	Enter the end time for the filter
Receivers	Enter the email recipient, while searching for multiple recipients, please separate them with a comma and no spaces.
Send result	Enter the status of the send result to filter with
Return code	Enter the email code to filter with
Email send module	Select the email module to filter from the drop-down list, which contains: All modules; User password; Alert events; CDR;

Table 23: Email Log Filter

SNMP

GXW450x supports SNMP (Simple Network Management Protocol) which is widely used in network management for network monitoring for colle information about monitored devices. To configure SNMP settings, go to GXW450x Web **GUI** → **System Settings** → **SNMP**.

Note

The SNMP page can be viewed by the Administrator level users.



System Settings

Network Settings

DDNS Settings

Time Settings

SNMP

*Contact Email:

Proxy:

Enable SNMP Trap

SNMP Trap Proxy

Listening Port:

Figure 44: SNMP Settings

This page has five tabs: SNMP Settings, SNMP Community, SNMP Trap Destination, SNMP V3 Users, and SNMP Trap Proxy. Please refer to the beliables for each tab.

SNMP Settings	
Enable	Enables SNMP feature. Default is Disabled .
Device Name	Configures the Device Name.
Device Location	Configures the Device Location.
Contact Email	Configures the email address of the administrator on which to receive notifications.
Enable SNMP Trap Proxy	Enables the SNMP Trap Proxy. Default is Disabled .
SNMP Trap Proxy Listening Port	Configures the SNMP Trap Proxy Listening Port. The default port is 162 .
SNMP Community	
Name	Community string associated with the trap. It must match the community string of the receiver.
Access Level	Configure the access level. Two levels are available: Read only: Can view the device configuration. Read/Write: Can view and change the device configuration.
SNMP Trap Destinations (GXW450x as managed device)	
Name	Configure the Name for the SNMP Trap Destination.

IP Address	The IP address of the SNMP trap receiver.
Port	Configure the SNMP trap receiver listening port.
Community	Community is by default set to Public, as community strings for SNMP v1 and v2 aren't encrypted.
	They are 3 available types:
Туре	Trapsink : to send SNMP v1 traps
	Trap2sink: to send SNMP v2 traps.
	Informsink: to send inform notification.
SNMP V3 Users	
Name	Configure the Name of the SNMP v3 Users.
Authentication Protocol	Authentication Protocol for SNMPv3. Available protocols are MD5 and SHA.
Authentication Password	Authentication Password for SNMPv3.
Privacy Protocol	Privacy protocol for SNMPv3. Available protocols are: DES, AES-128, AES-192, AES-256
Privacy Password	Privacy password for SNMPv3.
	Configure the group level, two levels are available:
Group Level	Read only: Can view the device configuration.
	Read/Write: Can view and change the device configuration.
SNMP Trap Proxy (GXW450x as Trap Proxy)	
Name	Configures the Name of the SNMP Trap Proxy.
Name	Note: "Enable SNMP Trap Proxy" needs to be toggled on.
IP Address	Configure the IP address of the SNMP manager.
Port	Configure the SNMP manager listening port.

Table 24: SNMP Parameters

TR-069

The GXW450x series supports TR-069 for remote management of equipment by service providers, reducing on-site visits and downtime. To configure SNMP settings, go to GXW450x Web $\textbf{GUI} \rightarrow \textbf{System Settings} \rightarrow \textbf{SNMP}$.

Note

The SNMP page can be viewed by the Administrator level users.

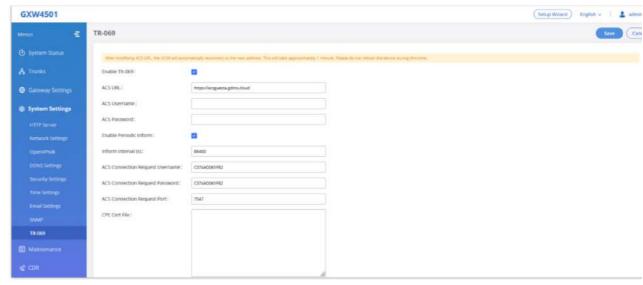


Figure 45: TR-069 Settings

Note: Reboot the device to make changes take effect. ACS URL Specifies URL of TR-069 ACS (e.g., https://acsguesta.gdms.cloud), or IP address. ACS Username Enters username to authenticate to ACS. ACS Password Enters password to authenticate to ACS. Enable Periodic Inform Sends periodic inform packets to ACS. The default is "No". Configures to send periodic "Inform" packets to ACS based on a specified intervals. The default setting is 86400. ACS Connection Request Username Enters username for the ACS to connect to the device. ACS Connection Request Password Enters the password for the ACS to connect to the device. CPE Cert File Uploads Cert File for the device to connect to the ACS via SSL.		
ACS Username Enters username to authenticate to ACS. ACS Password Enters password to authenticate to ACS. Enable Periodic Inform Sends periodic inform packets to ACS. The default is "No". Configures to send periodic "Inform" packets to ACS based on a specified intervals. The default setting is 86400. ACS Connection Request Username Enters username for the ACS to connect to the device. ACS Connection Request Password Enters the password for the ACS to connect to the device. CPE Cert File Uploads Cert File for the device to connect to the ACS via SSL.	Enable TR-069	
ACS Password Enable Periodic Inform Sends periodic inform packets to ACS. The default is "No". Configures to send periodic "Inform" packets to ACS based on a specified intervals. The default setting is 86400. ACS Connection Request Username Enters username for the ACS to connect to the device. ACS Connection Request Password Enters the password for the ACS to connect to the device. CPE Cert File Uploads Cert File for the device to connect to the ACS via SSL.	ACS URL	Specifies URL of TR-069 ACS (e.g, https://acsguesta.gdms.cloud), or IP address.
Enable Periodic Inform Sends periodic inform packets to ACS. The default is "No". Configures to send periodic "Inform" packets to ACS based on a specified intervals. The default setting is 86400. ACS Connection Request Username Enters username for the ACS to connect to the device. ACS Connection Request Password Enters the password for the ACS to connect to the device. CONNECTION Request Port Enters the port for the ACS to connect to the device. CONNECTION Request Port Uploads Cert File for the device to connect to the ACS via SSL.	ACS Username	Enters username to authenticate to ACS.
Configures to send periodic "Inform" packets to ACS based on a specified intervals. The default setting is 86400. ACS Connection Request Username Enters username for the ACS to connect to the device. ACS Connection Request Password Enters the password for the ACS to connect to the device. ACS Connection Request Port Enters the port for the ACS to connect to the device. CPE Cert File Uploads Cert File for the device to connect to the ACS via SSL.	ACS Password	Enters password to authenticate to ACS.
periodic "Inform" packets to ACS based on a specified intervals. The default setting is 86400. ACS Connection Request Username Enters username for the ACS to connect to the device. ACS Connection Request Password Enters the password for the ACS to connect to the device. ACS Connection Request Port Enters the port for the ACS to connect to the device. CPE Cert File Uploads Cert File for the device to connect to the ACS via SSL.	Enable Periodic Inform	Sends periodic inform packets to ACS. The default is "No".
ACS Connection Request Password Enters the password for the ACS to connect to the device. ACS Connection Request Port Enters the port for the ACS to connect to the device. CPE Cert File Uploads Cert File for the device to connect to the ACS via SSL.	Inform Interval (s)	periodic "Inform" packets to ACS based on a
ACS Connection Request Port Enters the port for the ACS to connect to the device. CPE Cert File Uploads Cert File for the device to connect to the ACS via SSL.	ACS Connection Request Username	Enters username for the ACS to connect to the device.
CPE Cert File Uploads Cert File for the device to connect to the ACS via SSL.	ACS Connection Request Password	Enters the password for the ACS to connect to the device.
	ACS Connection Request Port	Enters the port for the ACS to connect to the device.
CPE Cert Key Uploads Cert Key for the device to connect to the ACS via SSL.	CPE Cert File	Uploads Cert File for the device to connect to the ACS via SSL.
	CPE Cert Key	Uploads Cert Key for the device to connect to the ACS via SSL.

Table 25 : TR-069 Settings

TRUNKS

GXW450X is a VoIP Digital Gateway that supports both trunk modes Digital and VoIP to ensure a smooth integration of digital and VoIP communi to connect the legacy telephony infrastructure made up of PRI (E1, T1, J1) to the IP network.

Digital Trunks

The GXW450X supports E1/T1/J1 which are physical connection technologies used in the digital network. T1 is the North American standard, J1 is Japan, whereas E1 is the European standard. GXW450X supports four signaling protocols: PRI_NET, PRI_CPE, MFC/R2, and SS7. PRI provides a vary number of channels depending on the standards in the country of implementation (E1, T1, or J1); MFC/R2 is a signaling protocol heavily used ove trunks; SS7 uses out-of-band signaling, which travels on a separate, dedicated channel rather than within the same channel as the telephone call, providing more efficiency and higher security level when the telephone calls are set up.

To set up a digital trunk on the GXW450X:

- 1. Go to Web GUI→Gateway Settings→Interface Settings→Digital Hardware to configure port type and channels.
- 2. Go to Web GUI->Trunks->Digital Trunks to add and edit the digital trunks.
- 3. Go to Web GUI → Trunks→Outbound Routes and Inbound Routes to configure outbound and inbound rules for the digital trunk.

Digital Hardware Configuration

Go to Web GUI -> Gateway Settings->Interface Settings-> Digital Hardware page and configure the following:

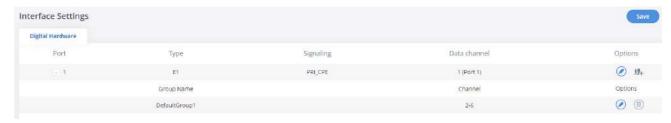


Figure 45: Digital Hardware Configuration

- o Click on o to edit digital ports. Please see configuration parameters in the tables below:
- Click on oto edit group. This assigns channels to be used for the digital port. For E1, 30 B channels can be assigned to the default group; fo 23 B channels can be assigned to the default group.
- If fewer than 30 B channels for E1 or 23 B channels for T1/J1 are assigned in the default group, users can click on to add more groups. This is necessary in most cases and only the default group is needed.

GXW4501

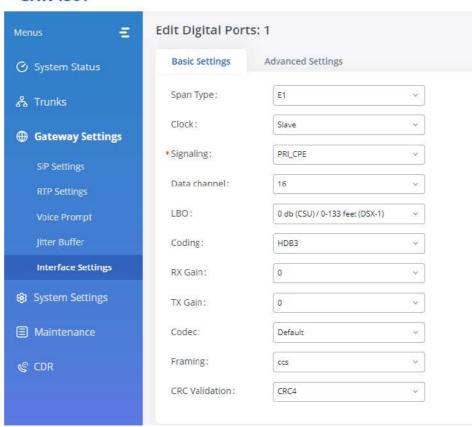


Figure 46: Digital Port Configuration

The GXW450X currently supports E1, T1, and J1 digital hardware types. When different signaling is selected for E1, T1, or J1, the settings in basic o and advanced options will be different. The following tables list all the settings to configure digital ports when selecting each signaling.

Basic Settings	
Clock	All E1/T1/J1 spans generate a clock signal on their transmit side. The parameter determines whether the clock signal fron far end of the E1/T1/J1 is used as the master source of clock timing. If the far end is used as the master, the gateway syste clock will synchronize to it. Master: The port will never be used as a source of timing. This is appropriate when you know the far end should always slave to you. Slave: The equipment at the far end of the E1/T1/J1 link is the preferred source of the master clock.

Signaling	Chooses the signaling protocol that will be used on the digital port, the available options are: PRI_NET, PRI_CPE, SS7, MFC/R2
Data channel	Chooses the Data Channel for control. E1: "AMI" or "HDB3"
LBO	The line build-out (LBO) is the distance between the operators and the gateway. Please use the default value of 0dB unles distance is long. T1: "AMI" or "B8ZS" E1: "AMI" or "HDB3"
Coding	T1: "AMI" or "B8ZS"
RX Gain	Configure the RX gain for the receiving channel of the digital port. The valid range is from -24dB to +12dB.
TX Gain	Configure the TX Gain for the transmitting channel of the digital port. The valid range is -24dB to +12dB.
Codec	Select alaw or ulaw. If set to default, alaw will be used for E1.
Framing	If the span type is E1, the signaling configured as MFC/R2, then framing must configure as "cas"; If the span type is E1, signaling configured as PRI or SS7, then framing must configure as "ccs"; If span type is T1, and the signaling configure PRI or SS7, then framing can configure as "esf" or "d4"; If span type is J1, and the signaling configured as PRI or SS7, then framing can configure as "esf" or "d4".
CRC Validation	For E1, select whether to use CRC4 or None.
Advanced Settings	
Switch Type	Select switch type. EuroISDN: EuroISDN (common in Europe) NI2: National ISDN type 2 (common in the US) DMS100: Nortel DMS100 4ESS: AT&T 4ESS 5ESS: Lucent 5ESS NI1: old national ISDN type 1 Q.SIG
PRI Dial Plan	This setting is used to specify the type of the callee number. The service provider will usually verify this. The default sett "unknown". In some very unusual circumstances, you may need set it to "Dynamic" or "Redundant". Note: When one type is selected, you might not be able to dial another class of numbers. For example, if "National" is configured, you won't be able to dial local or international numbers.
PRI Local Dial Plan	This setting is used to specify the type of caller number. The service provider will usually verify this.
International Prefix National Prefix Local Prefix Private Prefix Unknown Prefix	Configure the prefix in PRI Local Dial Plan and PRI Dial Plan for each type.
PRI T310	Configure PRI T310 Timer (in seconds). The default value is 10 seconds.
PRI Indication	Select the PRI Indication. outofband: Use RELEASE, DISCONNECT, or other messages with CAUSE to indicate call progress (e.g., cause: unassignumber or user busy). inband: use in-band tones to play busy or congestion signals to the other side. This is the default setting.
Reset Interval	The interval that restarts idle channels.
PRI Exclusive	This setting is used to set up the ChannelD in the SETUP message. If enabled, only the specified B channel can be used. Otherwise, select one of the channels in the B channel. If you need to override the existing channels selection routine and all PRI channels to be marked as exclusively selected, please enable it.

Facility Enable	If selected, the transmission of facility-based ISDN supplementary services (such as caller name from CPE over facility) be enabled.
SETUPACK	When receiving a remote "SETUP" SIP message, and the "Sending Complete" field is not included in it, the gateway wil a "SETUP ACK" to request more information. This option should be used if a remote device has "SETUP ACK" support issues.
Overlap Dial	Configure this option to send overlap digits. If enabled, the SETUP message can include some digits of the callee number the rest of the digits can be sent using the INFORMATION message. If disabled, the callee number will be sent via SETU message when all the digits are ready.
NSF	Some switches (AT&T especially) require network-specific facilities. Currently the supported values are "none", "sdn", "megacom", "tollfreemegacom".
PROGRESS	If enabled, GXW450x can send a signaling message to the calling party indicating that the call is still in progress and that called party has not yet answered. This can be helpful in situations where the call setup time is longer than expected, or w there may be delays in the network. If disabled, the pri incoming calls to GXW450x converts the PROGRESS message into ALERTING message and send it PRI trunk. This option is used to determine whether the peer supports the PROGRAMS message. Enabled By Default.

Table 25:Digital Hardware Configuration Parameters: E1 – PRI_NET/PRI_CPE

Basic Settings	
Clock	All E1/T1/J1 spans generate a clock signal on their transmit side. The parameter determines whether the clock signal from far end of the E1/T1/J1 is used as the master source of clock timing. If the far end is used as the master, the gateway syste clock will synchronize to it. Master: The port will never be used as a source of timing. This is appropriate when you know the far end should always slave to you. Slave: The equipment at the far end of the E1/T1 link is the preferred source of the master clock.
Signaling	Chooses the signaling protocol that will be used on the digital port, the available options are: PRI_NET, PRI_CPE, SS7, MFC/R2 PRI: when one end is set to NET, the other end should be set to CPE.
Data channel	The Data channel for control. Specifies the channel to use for data connections when PRI_NET or PRI_CPE is chosen as signaling protocol. While, the first dropdown list specifies the E1/T1 port to use, and the second specifies the channel to the for data connections when SS7 is chosen. The user can group multiple E1 lines with a single data channel.
SS7 Variant	Select ITU, ANSI, or CHINA.
Originating Point Code	Originating point code is used to identify the node originating the message, always provided by the operator/ISP. ITU Format: decimal number. ANSI & CHINA Format: decimal number or XXX-XXX-XXX.
Destination Point Code	The destination point code is the address to send the message to, always provided by the operator/ISP. ITU Format: decimal number. ANSI & CHINA Format: decimal number or XXX-XXX-XXX.
First CIC	When Span Type is E1, ITU & CHINA Range: [0, 4065], ANSI Range: [0, 16353]. When Span Type is T1/J1, ITU & CHINA Range: [0, 4072], ANSI Range: [0, 16360].
Assign CIC To D-channel	If set to yes, D-channel will be assigned a CIC. Else, D-channel will not be assigned. By default, it is set to No.
Network Indicator	Network Indicator (NI) should match in nodes, otherwise, it might cause issues. Users can select "National", "National S ₁ "International", or "International Spare". Usually, "National" or "International" is used.
LBO	The line build-out (LBO) is the distance between the operators and the gateway. Please use the default value of 0dB unles distance is long.
Coding	T1:"AMI" or "B8ZS" And E1:"AMI" or "HDB3"
RX Gain	Configure the RX gain for the receiving channel of the digital port. The valid range is from -24dB to +12dB.

TX Gain	Configure the TX Gain for the transmitting channel of the digital port. The valid range is -24dB to +12dB.
Codec	Select alaw or ulaw. If set to default, alaw will be used for E1.
Framing	If the span type is E1, the signaling configured as MFC/R2, then framing must configure as "cas"; If the span type is E1, the signaling configured as PRI or SS7, then framing must configure as "ccs"; If span type is T1, and the signaling configured as PRI or SS7, then framing can configure as "esf" or "d4"; If span type is J1, and the signaling configured as PRI or SS7, then framing can configure as "esf" or "d4".
CRC Validation	For E1, select whether to use CRC4 or None.
Advanced Settings	
Called Nature of Address Indicator	Indicates the type of the called number. The receiving switch may use this indicator during translations to apply the numb proper dial plan. Users can select "Unknown", "Subscriber", "National", "International" or "Dynamic".
Calling Nature of Address Indicator	Indicates the type of the calling number. The receiving switch may use this indicator during translations to apply the num proper dial plan. Users can select "Unknown", "Subscriber", "National", "International" or "Dynamic".
Orginal Called	This option decides on SS7 trunk outgoing calls,By controlling the "original called number IE" (Information Element) in signaling messages, the user can ensure that the correct phone number is displayed to the recipient of a call, even when th has been rerouted or redirected. Example: if a call is routed through multiple networks or carriers, the "original called number" information can be lost or modified along the way. However, by using this SS7 option, the user can preserve the original called number and ensure t is displayed correctly to the recipient. Disabled by Default.
Early ACM	Early ACM can be used to provide immediate feedback to callers that their call is being connected, as opposed to hearing silence or ringing until the call is actually connected to the intended party. This can help to reduce perceived wait times at improve the overall user experience. If enabled, When an inbound call is received by the gateway, the gateway can signal to the calling party that the call is be connected and that the called party will begin ringing. This early answer supervision signal is sent by the gateway before called party's phone rings. Disabled by Default.
International Prefix National Prefix Subscriber Prefix Unknown Prefix	Configure the prefix in Called Nature of Address Indicator and Calling Nature of Address Indicator for each type.

Digital Hardware Configuration Parameters: E1 – SS7

Basic Settings	
Clock	All E1/T1/J1 spans generate a clock signal on their transmit side. The parameter determines whether the clock signal from the f end of the E1/T1/J1 is used as the master source of clock timing. If the far end is used as the master, the gateway system clock synchronize to it. • Master: The port will never be used as a source of timing. This is appropriate when you know the far end should always be slave to you. • Slave: The equipment at the far end of the E1/T1 link is the preferred source of the master clock.
Signaling	Chooses the signaling protocol that will be used on the digital port. PRI: when one end is set to NET, the other end should be set to CPE
Data channel	Chooses the Data Channel for control. The user can group multiple E1 lines with a single data channel.
Variant	MFC/R2 multinational adaption. GXW450X supports MFC/R2 standards by ITU and MFC/R2 standards in different countries or regions including Argentina, Brazil, China, Czech Republic, Colombia, Ecuador, Indonesia, Mexico, the Philippines, and Venezue

Category	Defines the Caller Category. Users can choose among the following options: National Subscriber, National Priority Subscriber, International Subscriber, and International Priority Subscriber.
Get ANI First	If enabled, the callee side will request the caller to send the caller number first and then called number.
	Note: Options "Get ANI First" and "Skip Category" cannot be enabled at the same time.
LBO	The line build-out (LBO) is the distance between the operators and the gateway. Please use the default value of 0dB unless the distance is long.
Coding	T1: "AMI" or "B8ZS"
coug	E1: "AMI" or "HDB3"
RX Gain	Configure the RX gain for the receiving channel of the digital port. The valid range is from -24dB to +12dB.
TX Gain	Configure the TX Gain for the transmitting channel of the digital port. The valid range is -24dB to +12dB.
	If the span type is E1, the signaling configured as MFC/R2, then framing must configure as "cas";
Framing	If the span type is E1, the signaling configured as PRI or SS7, then framing must configure as "ccs";
J	If the span type is T1, and the signaling configured as PRI or SS7, then framing can configure as "esf" or "d4";
	If span type is J1, and the signaling configured as PRI or SS7, then framing can configure as "esf" or "d4".
CRC Validation	For E1, select whether to use CRC4 or None.
Advanced Settings	
MF Back Timeout (ms)	MFC/R2 value in milliseconds for MF timeout. Values smaller than 500ms are not recommended1 represents the default value
Metering Pulse Timeout	MFC/R2 value in milliseconds for the metering pulse timeout. Metering pulse is sent by some telcos for some R2 variants durin call presumably for billing purposes to indicate costs. Should not last more than 500ms, -1 represents the
(ms)	default value, and for Argentina, the default value is 400ms, for others is 0ms.
Allow Collect	Brazil has a special calling party category for collect calls (llamadas por cobrar) instead of using the operator (as in Mexico). The spec in Brazil says a special GB tone should be used to reject collect calls.
Cans	By default, this is disabled, which means collect calls will be blocked.
Double Answer	Some gateways require a double-answer process to block collect calls. If users have a problem blocking collect calls using Grousignals, please try enabling this option.
Accept On	By default, it's enabled. In most
Offer	the cases, this option should be enabled.
Skin Catagory	If enabled, the callee side will request the caller to send the caller category before sending the caller number.
Skip Category	Note: "Get ANI First" and "Skip Category" cannot be enabled at the same time.
	Whether or not to report to the other end "accept call with charge". This setting has no effect on most telecoms. The default se

Custom Options

Click on the "Custom Options" button (on the left top of the configuration dialog) and then the user can customize desired ton timer options accordingly.

Table 27: Digital Hardware Configuration Parameters: E1 – MFC/R2

Basic Settings	
Clock	All E1/T1/J1 spans generate a clock signal on their transmit side. The parameter determines whether the clock signal fron far end of the E1/T1/J1 is used as the master source of clock timing. If the far end is used as the master, the gateway syste clock will synchronize to it. Master: The port will never be used as a source of timing. This is appropriate when you know the far end should always slave to you. Slave: The equipment at the far end of the E1/T1/J1 link is the preferred source of the master clock.
Signaling	Chooses the signaling protocol that will be used on the digital port. PRI: when one end is set to NET, the other end should be set to CPE.
Data channel	Chooses the Data Channel for control. The user can group multiple E1 lines with a single data channel.
LBO	The line build-out (LBO) is the distance between the operators and the gateway. Please use the default value of 0dB unles distance is long.
Coding	T1: "AMI" or "B8ZS" E1: "AMI" or "HDB3"
RX Gain	Configure the RX gain for the receiving channel of the digital port. The valid range is from -24dB to +12dB.
TX Gain	Configure the TX Gain for the transmitting channel of the digital port. The valid range is -24dB to +12dB.
Codec	Select alaw or ulaw. If set to default, ulaw will be used for T1/J1.
Framing	Select "esf" or "d4". The default setting is esf.
Advanced Settings	
Switch Type	Select switch type. EuroISDN: EuroISDN (common in Europe) NI2: National ISDN type 2 (common in the US) DMS100: Nortel DMS100 4ESS: AT&T 4ESS 5ESS: Lucent 5ESS NI1: old national ISDN type 1 Q.SIG
PRI Dial Plan	This setting is used to specify the type of the callee number. The service provider will usually verify this. The default sett "unknown". In some very unusual circumstances, you may need set it to "Dynamic" or "Redundant". Note: When one type is selected, you might not be able to dial another class of numbers. For example, if "National" is configured, you won't be able to dial local or international numbers.
PRI Local Dial Plan	This setting is used to specify the type of caller number. The service provider will usually verify this.
International Prefix National Prefix Local Prefix Private Prefix Unknown Prefix	Configure the prefix in PRI Local Dial Plan and PRI Dial Plan for each type.
PRI T310	Configure PRI T310 Timer (in seconds). The default value is 10 seconds.
PRI Indication	Select the PRI Indication. outofband: Use RELEASE, DISCONNECT, or other messages with CAUSE to indicate call progress (e.g., cause: unassiţ number or user busy).

	inband: use in-band tones to play busy or congestion signals to the other side. This is the default setting. The interval that restarts idle channels.
Reset Interval	The interval that restarts idle channels.
PRI Exclusive	This setting is used to set up the ChannelID in the SETUP message. If enabled, only the specified B channel can be used. Otherwise, select one of the channels in the B channel. If you need to override the existing channels selection routine and all PRI channels to be marked as exclusively selected, please enable it.
Facility Enable	If selected, the transmission of facility-based ISDN supplementary services (such as caller name from CPE over facility) be enabled.
SETUPACK	When receiving a remote "SETUP" SIP message, and the "Sending Complete" field is not included in it, the gateway wil a "SETUP ACK" to request more information. This option should be used if a remote device has "SETUP ACK" support issues.
Overlap Dial	Configure this option to send overlap digits. If enabled, the SETUP message can include some digits of the callee number the rest of the digits can be sent using the INFORMATION message. If disabled, the callee number will be sent via SETU message when all the digits are ready.
NSF	Some switches (AT&T especially) require network-specific facilities. Currently the supported values are "none", "sdn", "megacom", "tollfreemegacom", "accunet".
PROGRESS	If disabled, the pri incoming calls GXW450X to convert the PROGRESS message into ALERTING message and send it PRI trunk. This option is used to determine whether the peer supports the PROGRAMS message. Enabled by Default.

Table 28: Digital Hardware Configuration Parameters: T1/J1 – PRI_NET/PRI_CPE

	Table 28: Digital Hardware Configuration Parameters: 11/J1 – PRI_NE1/PRI_CPE		
Basic Settings			
Clock	All E1/T1/J1 spans generate a clock signal on their transmit side. The parameter determines whether the clock signal from the end of the E1/T1/J1 is used as the master source of clock timing. If the far end is used as the master, the gateway system clowill synchronize to it. • Master: The port will never be used as a source of timing. This is appropriate when you know the far end should always slave to you. • Slave: The equipment at the far end of the E1/T1 link is the preferred source of the master clock.		
Signaling	Chooses the signaling protocol that will be used on the digital port. PRI: when one end is set to NET, the other end should be set to CPE		
Data channel	Chooses the Data Channel for control. The user can group multiple E1 lines with a single data channel.		
SS7 Variant	Select ITU, ANSI, or CHINA.		
Originating Point Code	Originating point code is used to identify the node originating the message, always provided by the operator/ISP. ITU Format: decimal number. ANSI & CHINA Format: decimal number or XXX-XXX.		
Destination Point Code	The destination point code is the address to send the message to, always provided by the operator/ISP. o ITU Format: decimal number. o ANSI & CHINA Format: decimal number or XXX-XXX-XXX.		
First CIC	When Span Type is E1, ITU & CHINA Range: [0, 4065], ANSI Range: [0, 16353]. When Span Type is T1/J1, ITU & CHINA Range: [0,4072], ANSI Range: [0, 16360].		

If set to yes, D-channel will be assigned with a CIC. Else, D-channel will not be assigned with a CIC. By default, it is set to No	
Network Indicator (NI) should match in nodes, otherwise, it might cause issues. Users can select "National", "National Spare". Usually, "National" or "International" is used.	
The line build-out (LBO) is the distance between the operators and the gateway. Please use the default value of 0dB unless distance is long.	
T1: "AMI" or "B8ZS" E1: "AMI" or "HDB3"	
Configure the RX gain for the receiving channel of the digital port. The valid range is from -24dB to +12dB.	
Configure the TX Gain for the transmitting channel of the digital port. The valid range is -24dB to +12dB.	
Select alaw or ulaw. If set to default, ulaw will be used for T1/J1.	
Select "esf" or "d4". The default setting is esf.	
Indicates the type of the called number. The receiving switch may use this indicator during translations to apply the number proper dial plan. Users can select "Unknown", "Subscriber", "National", "International" or "Dynamic".	
Indicates the type of the calling number. The receiving switch may use this indicator during translations to apply the numproper dial plan. Users can select "Unknown", "Subscriber", "National", "International" or "Dynamic".	
Configure the prefix in Called Nature of Address Indicator and Calling Nature of Address Indicator for each type.	

Table 29: Digital Hardware Configuration Parameters: T1/J1 – SS7

Basic Settings	
Clock	All E1/T1/J1 spans generate a clock signal on their transmit side. The parameter determines whether the clock signal from the far er the E1/T1/J1 is used as the master source of clock timing. If the far end is used as the master, the gateway system clock will synchro to it. • Master: The port will never be used as a source of timing. This is appropriate when you know the far end should always be a sla you. • Slave: The equipment at the far end of the E1/T1 link is the preferred source of the master clock.
Signaling	Chooses the signaling protocol that will be used on the digital port.
Coding	T1: "AMI" or "B8ZS"
RX Gain	Configure the RX gain for the receiving channel of the digital port. The valid range is from -24dB to +12dB.
TX Gain	Configure the TX Gain for the transmitting channel of the digital port. The valid range is -24dB to +12dB.

Codec	Select alaw or ulaw. The default codec is "ulaw" for T1.	
Framing	Select "esf" or "d4". The default setting is esf.	
Advanced Settings		
RX Wink	Sets the receive wink timing. Default settings is 300ms.	

Table 30: Digital Hardware Configuration Parameters: T1 – E&M Immediate

Digital Trunk Configuration

After configuring digital hardware, go to Web GUI→ Trunks→Digital Trunks.

- Click on + Create New Digital Trunk to add a new digital trunk.
- o Click on o to configure detailed parameters for the digital trunk.
- o Click on to delete the digital trunk.

The digital trunk parameters are listed in the table below.

Trunk Name	Configure trunk name to identify the digital trunk.	
Port	Configure the digital channel group used by the trunk.	
Hide CallerID	Configure to hide outgoing caller ID. The default setting is "No".	
Caller ID	Configure the Caller ID. This is the number that the trunk will try to use when making outbound calls. For some providers, it mi not be possible to set the CallerID with this option and this option will be ignored.	
CallerID Name	Configure the name of the caller.	
DAHDI Out Line Selection	 This is to implement a Digital trunk outbound line selection strategy. Three options are available: Ascend: When the call goes out from this digital trunk, it will always try to use the first idle digital port. The port order that call will use to go out would be port 1→port 2→port 3→port 4. Every time it will start with port 1 (if it's idle). Poll: When the call goes out from this digital trunk, it will use the port that is not used last time. And it will always use the in the order of port 1→2→3→4→1→2→3→4→1→2→3→4, following the last port being used. Descend: When the call goes out from this digital trunk, it will always try to use the last idle digital port. The port order that call will use to go out would be port 16→port 10→port 2→port 1. Every time it will start with port 4 (if it's idle). The default setting is "Ascend" mode. 	
Fax Gateway	Enable/disable Fax Gateway on the digital trunk. If enabled, GXW450X will detect the fax tone on the digital interface in order to initiate T.38 fax, otherwise, it will be sent in auc pass-through.	

Table 31: Digital Trunk Configuration Parameters

Digital Trunk Troubleshooting

After configuring the digital trunk on the GXW450X as described above, if it doesn't work as expected, users can go to capture the signaling trace GXW450X Web GUI for troubleshooting purposes.

Depending on the signaling selected for the digital trunk, users can go to the following pages to capture a trace:

PRI Signaling Trace: Web GUI→Maintenance→Signaling Troubleshooting→PRI Signaling Trace

SS7 Signaling Trace: Web GUI→Maintenance→Signaling Troubleshooting→SS7 Signaling Trace

MFC/R2 Signaling Trace: Web GUI→Maintenance→Signaling Troubleshooting→MFC/R2 Signaling Trace

Users can also capture a Digital Record Trace to record the call for other troubleshooting purposes such as audio quality problems and noise.

Below are the steps to capture the trace:

- 1. Click on "Start" to start capturing traces. The output result shows "Capturing..."
- 2. Once the test is done, click on "Stop" to stop the trace.
- 3. Click on "Download" to download the trace.



Figure 47: Troubleshooting Digital Trunks

After capturing the trace, users can download it for basic analysis. Or you can contact Grandstream Technical support in the following link for furth assistance if the issue is not resolved:

https://www.grandstream.com/support

VoIP Trunks

The VoIP trunks allow the GXW450X to be connected over an IP network via SIP protocol to a VoIP provider or to another device that supports the trunks.

VoIP trunks can be configured in GXW450X under Web GUI > **Trunks** > **VoIP Trunks**. Once created, the VoIP trunks will be listed with Provider Nar Type, Hostname/IP, Username, and Options to edit/detect the trunk.

Click on + Add SIP Trunk to add a new VoIP trunk.
Click on to configure detailed parameters for the VoIP trunk.
Click on to delete the VoIP trunk.
Click on to configure DOD.

The VoIP trunk options are listed in the table below.

	Select VoIP trunk type to create.	
Туре	o Peer SIP Trunk	
	Register SIP Trunk	
Provider Name	Configure a unique label (up to 64 characters) to identify this trunk when listed in outbound rules, inbound rules and etc.	
Host Name	Configure the IP address or URL for the VoIP provider's server of the trunk.	
NAT	Turn on this setting when the gateway is using public IP and communicating with devices behind NAT. If there is a one-way a issue, usually it is related to NAT configuration or SIP/RTP port support on the firewall.	
Disable This Trunk	If checked, the trunk will be disabled.	
	Note: If a current "Register SIP trunk" is disabled, GXW450X will send UNREGISTER message (REGISTER message with expires to the SIP provider.	
TEL URI	If the trunk has an assigned PSTN telephone number, this field should be set to "User=Phone". Then a "User=Phone" param will be attached to the Request-Line and TO header in the SIP request to indicate the E.164 number. If set to "Enable", "Tel:" used instead of "SIP:" in the SIP request. The default setting is disabled.	
From Domain	Configure the actual domain name. This can be used to override the "From" Header.	
Domail	For example, "trunk.GXW450X.provider.com" is the From Domain in From Header: sip: 1234567@trunk.GXW450X.provider.co	

Transport	Configure the SIP transport protocol to be used in this trunk. UDP; TCP or TLS. The default setting is "UDP"	
If "Type" is set to "Register SIP Trunk"		
Need Registration	Select whether the trunk needs to register on the external server or not when the "Register SIP Trunk" type is selected. The disetting is "Yes".	
Allow outgoing calls if registration fails	If enabled outgoing calls even if the registration to this trunk fails will still be able to go through. Note that if we uncheck the "Need Registration" option, this option will be ignored. The default setting is "Yes".	
Username	Enter the username to register to the trunk from the provider.	
Password	Enter the password to register to the trunk from the provider.	
AuthID	Enter the Authentication ID to register to the trunk from the provider.	

Table 32: Create New SIP Trunk

After creating the SIP Trunk user can click on 🕖 to edit the trunk and have detailed parameters to configure. Below is a table of the Basic and ad parameters of a SIP trunk.

Basic Settings	
Provider Name	Configure a unique label to identify this trunk when listed in outbound rules, inbound rules and etc.
Host Name	Configure the IP address or URL for the VoIP provider's server of the trunk.
Secondary SIP Server	The URL or IP address, and port of the SIP server. This will be used when the primary SIP server fails.
Keep Original CID	Keep CID from the inbound call when dialing out even if the option "Keep Trunk CID" is enabled. Please make sure the GXW at the other end supports matching user entry using the "username" field from the authentication line.
Keep Trunk CID	Keep trunk CID configured on Basic settings.
NAT	Turn on this option when the gateway is using public IP and communicating with devices behind NAT. If there is a one-waudio issue, usually it's related to NAT configuration or SIP/RTP port configuration on the firewall.
Disable This Trunk	If selected, the trunk will be disabled. Note: If a current SIP trunk is disabled, GXW450X will send UNREGISTER message (REGISTER message with expire to the SIP provider.
TELURI	If the trunk has an assigned PSTN telephone number, this field should be set to "User=Phone". Then a "User=Phone" parameter will be attached to the Request-Line and TO header in the SIP request to indicate the E.164 number. If set to "Enable", "Tel:" will be used instead of "SIP:" in the SIP request. The default setting is disabled.
CallerID Name	Configure the new name of the caller when the extension has no CallerID Name configured.
From Domain	Configure the actual domain name. This can be used to override the "From" Header. For example, "trunk.GXW450X.provider.com" is the From Domain in From Header: sip:1234567@trunk.GXW450X.provider.com.
Transport	Configure the SIP transport protocol to be used in this trunk. 1. UDP 2. TCP 3. TLS The default setting is "UDP".
Need Registration	Select whether the trunk needs to register on the external server or not when the "Register SIP Trunk" type is selected. The

	default setting is "Yes".
Allow outgoing calls if registration fails	If enabled outgoing calls even if the registration to this trunk fails will still be able to go through. Note that if we uncheck "Need Registration" option, this option will be ignored. The default setting is "Yes".
From User	Configures the actual username of the extension. This can be used to override the "From" Header
Username	Enter the username to register to the trunk from the provider.
Password	Enter the password to register to the trunk from the provider.
AuthID	Enter the Authentication ID to register to the trunk from the provider.
Advanced Settings	
Codec Preference	Select the audio codec for the VoIP trunk. The available codecs are: PCMU, PCMA, G.726, G.729, iLBC, G.722, AAL2-G.726-32, G.723, OPUS
Send PPI Header	If checked, the invite message sent to trunks will contain PPI (P-Preferred-Identity) Header.
PPI Mode	Configure how to set the PPI number, there are three possible options: 1. Default: Use the register number of the trunk. 2. Original CID: Use the original CID in the PPI header, if no original CID, use the default number. 3. DOD number: Use the DOD number in the PPI header, if no DOD number, use the default number.
Send PAI Header	If checked, the INVITE message sent from the trunk will contain PAI (P-Asserted-Identity) Header. Default is unchecked
PAI Header	The user and name of the PAI header. It is formatted as "name" or "" or "number"; if null, use the CID according to the priority.
DOD as From Name	If enabled and "From User" is configured, the INVITE's From header will contain the DOD number.
Passthrough PAI Header	If enabled and "Send PAI Header" is disabled, PAI headers will be preserved as calls pass through the GXW450X.
Outbound Proxy Support	Enable sending an outbound signal to the proxy instead of the devices directly. The default setting is "unchecked".
Outbound Proxy	When configured, the outbound signal will be sent to the proxy instead of the devices directly. The outbound proxy can b domain name or IP address.
Backup Outbound Proxy	Secondary Outbound Proxy will be used when the primary proxy cannot be connected.
Remove OBP from Route	If enabled, the Route header will be removed from SIP requests. The default setting is "No".
DID Mode	Configure where to get the destination ID of an incoming SIP call, from SIP Request-line or To-header. The default is set "Request-line".
DTMF Mode	Configure the default DTMF mode when sending DTMF on this trunk. Default: The global setting of DTMF mode will be used. The global setting for the DTMF Mode setting is under Web GUI→Gateway Settings→SIP Settings→ToS. RFC2833: Send DTMF using RFC2833. Info: Send DTMF using the SIP INFO message. Inband: Send DTMF using inband audio. This requires a 64-bit codec, i.e., PCMU and PCMA. Auto: Send DTMF using RFC2833 if offered. Otherwise, inband.
Enable Heartbeat Detection	If enabled, the GXW450X will regularly send SIP OPTIONS to the device to check if the device is still online. The defau setting is "No".
Heartbeat Frequency	When the "Enable Heartbeat Detection" option is set to "Yes", configure the interval (in seconds) of the SIP OPTIONS message sent to the device to check if the device is still online. The default setting is 60 seconds.
Maximum Number of Call Lines	The maximum number of concurrent calls using the trunk. The default setting 0, which means no limit.

SRTP	Enable SRTP for the VoIP trunk to use 1. Disabled 2. Enabled but not forced 3. Enabled and forced" The default setting is "Disabled". It uses SDP Security Description to exchange keys. Please refer to: SDES: https://tools.ietf.org/html/rfc4568 SRTP: https://www.ietf.org/rfc/rfc3711.txt
E1/T1/J1 Error Code	Selects the SIP response code to send to the VoIP trunk when the E1/T1/J1 interface is down or unavailable, the avilable vare 480 and 503, set to 480 by Default • Error Code 480: the requested service is temporarily unavailable • Error Code 503: the service is unavailable due to a server overload or maintenance

Table 33 : VoIP Trunk Configuration Parameters – Register SIP Trunk

Basic Settings	
Provider Name	Configure a unique label to identify this trunk when listed in outbound rules, inbound rules and etc.
Host Name	Configure the IP address or URL for the VoIP provider's server of the trunk.
Keep Original CID	Keep CID from the inbound call when dialing out even if the option "Keep Trunk CID" is enabled. Please make sure the GXW at the other end supports matching user entry using the "username" field from the authentication line.
Keep Trunk CID	Keep trunk CID configured on Basic settings.
NAT	Turn on this option when the gateway is using public IP and communicating with devices behind NAT. If there is a one-vaudio issue, usually it's related to NAT configuration or SIP/RTP port configuration on the firewall.
Disable This Trunk	If selected, the trunk will be disabled. Note: If a current SIP trunk is disabled, GXW450X will send UNREGISTER message (REGISTER message with expire to the SIP provider.
TELURI	If the trunk has an assigned PSTN telephone number, this field should be set to "User=Phone". Then a "User=Phone" parameter will be attached to the Request-Line and TO header in the SIP request to indicate the E.164 number. If set to "Enable", "Tel:" will be used instead of "SIP:" in the SIP request. The default setting is disabled.
Caller ID	Configure the Caller ID. This is the number that the trunk will try to use when making outbound calls. For some provide might not be possible to set the CallerID with this option and this option will be ignored.
CallerID Name	Configure the new name of the caller when the extension has no CallerID Name configured.
From Domain	Configure the actual domain name. This can be used to override the "From" Header. For example, "trunk.GXW450X.provider.com" is the From Domain in From Header: sip:1234567@trunk.GXW450X.provider.com.
Transport	Configure the SIP transport protocol to be used in this trunk. 1. UDP 2. TCP 3. TLS The default setting is "UDP".
Need Registration	Select whether the trunk needs to register on the external server or not when the "Register SIP Trunk" type is selected. The default setting is "Yes".
Allow outgoing calls if registration fails	If enabled outgoing calls even if the registration to this trunk fails will still be able to go through. Note that if we uncheck "Need Registration" option, this option will be ignored. The default setting is "Yes".
From User	Configures the actual username of the extension. This can be used to override the "From" Header

Username	Enter the username to register to the trunk from the provider.
Password	Enter the password to register to the trunk from the provider.
AuthID	Enter the Authentication ID to register to the trunk from the provider.
Advanced Settings	
Codec Preference	Select the audio codec for the VoIP trunk. The available codecs are: PCMU, PCMA, G.726, G.729, iLBC, G.722, AAL2-G.726-32, G.723, OPUS
Send PPI Header	If checked, the invite message sent to trunks will contain PPI (P-Preferred-Identity) Header.
PPI Mode	Configure how to set the PPI number, there are three possible options: 1. Default: Use the register number of the trunk. 2. Original CID: Use the original CID in the PPI header, if no original CID, use the default number. 3. DOD number: Use the DOD number in the PPI header, if no DOD number, use the default number.
Send PAI Header	If checked, the INVITE message sent from the trunk will contain PAI (P-Asserted-Identity) Header. Default is unchecked
PAI Header	The user and name of the PAI header. It is formatted as "name" or "" or "number"; if null, use the CID according to the priority.
DOD as From Name	If enabled and "From User" is configured, the INVITE's From header will contain the DOD number.
Passthrough PAI Header	If enabled and "Send PAI Header" is disabled, PAI headers will be preserved as calls pass through the GXW450X.
Outbound Proxy Support	Enable sending an outbound signal to the proxy instead of the devices directly. The default setting is "unchecked".
Outbound Proxy	When configured, the outbound signal will be sent to the proxy instead of the devices directly. The outbound proxy can be domain name or IP address.
Backup Outbound Proxy	Secondary Outbound Proxy will be used when the primary proxy cannot be connected.
Remove OBP from Route	If enabled, the Route header will be removed from SIP requests. The default setting is "No".
DID Mode	Configure where to get the destination ID of an incoming SIP call, from SIP Request-line or To-header. The default is set "Request-line".
DTMF Mode	Configure the default DTMF mode when sending DTMF on this trunk. Default: The global setting of DTMF mode will be used. The global setting for the DTMF Mode setting is under Web GUI→Gateway Settings→SIP Settings→ToS. RFC2833: Send DTMF using RFC2833. Info: Send DTMF using the SIP INFO message. Inband: Send DTMF using inband audio. This requires a 64-bit codec, i.e., PCMU and PCMA. Auto: Send DTMF using RFC2833 if offered. Otherwise, inband.
Enable Heartbeat Detection	If enabled, the GXW450X will regularly send SIP OPTIONS to the device to check if the device is still online. The defau setting is "No".
Heartbeat Frequency	When the "Enable Heartbeat Detection" option is set to "Yes", configure the interval (in seconds) of the SIP OPTIONS message sent to the device to check if the device is still online. The default setting is 60 seconds.
Maximum Number of Call Lines	The maximum number of concurrent calls using the trunk. The default setting 0, which means no limit.
SRTP	Enable SRTP for the VoIP trunk to use 1. Disabled 2. Enabled but not forced 3. Enabled and forced"

	The default setting is "Disabled". It uses SDP Security Description to exchange keys. Please refer to: SDES: https://tools.ietf.org/html/rfc4568 SRTP: https://www.ietf.org/rfc/rfc3711.txt
E1/T1/J1 Error Code	Selects the SIP response code to send to the VoIP trunk when the E1/T1/J1 interface is down or unavailable, the avilable vare 480 and 503, set to 480 by Default • Error Code 480: the requested service is temporarily unavailable • Error Code 503: the service is unavailable due to a server overload or maintenance

Table 34: VoIP Trunk Configuration Parameters – Peer SIP Trunk

Direct Outward Dialing (DOD)

The GXW450X provides Direct Outward Dialing (DOD) for both Digital and SIP trunks, which is a service of a local phone company (or local excharcarrier) that allows subscribers to connect to outside lines directly.

Example of how DOD is used:

Company ABC has a SIP trunk. This SIP trunk has 4 DIDs associated with it. The main number of the office is routed to an auto attendant. The othe numbers are direct lines to specific users of the company. Now when a user makes an outbound call their caller ID shows up as the main office nu This poses a problem as the CEO would like his calls to come from their direct line. This can be accomplished by configuring DOD for the CEO's number/extension.

Steps to configure DOD on the GXW4500:

- 1. To set up DOD go to GXW450X **Web GUI→Trunk→VoIP Trunks/Digital trunk** page.
- 2. Click to access the DOD options for the selected SIP Trunk.
- 3. Click "Create a new DOD" to begin your DOD setup
- 4. For "DOD Number" enter one of the numbers (DIDs) from your SIP/Digital trunk provider. In the example above

Company ABC received 4 DIDs from their provider. ABC will enter the number for the CEO's direct line.

- 5. Set the DOD name and If the extension number needs to be appended to the DID number click on "Add Extension".
- 6. Enter a number in the "Number" field. Users have the option of entering more than one number/extension separating them using ",". In this cas Company ABC would enter the CEO's numbers/extensions.

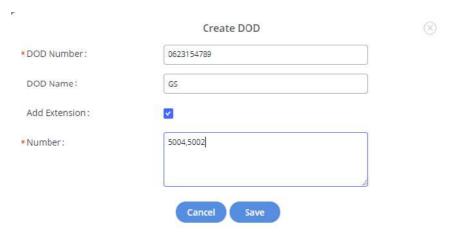


Figure 48: DOD configuration

7. Click "Save" at the bottom.

Once completed, the user will return to the EDIT DOD page which shows all the extensions that are associated with a particular DOD.



Figure 49: Edit DOD

Outbound Routes

An outbound route is a set of rules defined by privileges and patterns that the gateway uses to decide the numbers that can go out through the t and who has the right to use the trunk and trunk to use for an outbound call.

To create an outbound route, Go to Web GUI→ Trunks→Outbound Routes.

- Click on oto edit the outbound route.
- o Click on to delete the outbound route.

On the GXW450X, the outbound route priority is based on the "Best matching pattern". For example, the GXW450X has outbound route A with pa 1xxx and outbound route B with pattern 10xx configured. When dialing 1000 for an outbound call, outbound route B will always be used first. This because pattern 10xx is a better match than pattern 1xxx. Only when there are multiple outbound routes with the same pattern configured, the GXW450X will use the first pattern matched.



Figure 50: Create Outbound Route



	 All patterns are prefixed with the "_". Special characters:
	X: Any Digit from 0-9.
	Z : Any Digit from 1-9.
	N : Any Digit from 2-9.
	".": Wildcard. Match one or more characters.
D. 11	"!": Wildcard. Match zero or more characters immediately.
Pattern	Example: [12345–9] – Any digit from 1 to 9.
	Notes:
	Multiple patterns can be used. Each pattern should be entered in new line.
	o Example:
	_X.
	_NNXXNXXXX
	_818X.
	When enabled, users could specify extensions allowed to use this outbound route. "Privilege Level" is automatically disabled if "Enable Filter on Source Caller ID".
Enable Filter	The following two methods can be used at the same time to define the extensions as the source caller ID.
on Source Caller ID	1. Select available extensions/extension groups from the left to the right. This allows users to specify arbitrary single extensio available in the PBX.
	2. Custom Dynamic Route: define the pattern for the source caller ID. This allows users to define extension range instead of selecting them one by one.
Main Trunk	
Trunk	Select the trunk for this outbound rule.
	Allows the user to specify the number of digits that will be stripped from the beginning of the dialed string before the call is pl via the selected trunk.
Strip	Example:
	The users will dial 9 as the first digit of long-distance calls. However, 9 should not be sent out via digital lines and the PSTN line this case, 1 digit should be stripped before the call is placed.
Prepend	Specify the digits to be prepended before the call is placed via the trunk. Those digits will be prepended after the dialing number stripped.
Failover Trunk	
	Failover trunks can be used to make sure that a call goes through an alternate route when the primary trunk is busy or down. It Failover Trunk" is enabled and "Failover trunk" is defined, the calls that cannot be placed via the regular trunk may have a second trunk to go through.
Trunk	GXW450X supports up to 10 failover trunks.
	Example: The user's primary trunk is a VoIP trunk and the user would like to use the PSTN when the VoIP trunk is not available. PSTN trunk can be configured as the failover trunk of the VoIP trunk.

Strip	Allows the user to specify the number of digits that will be stripped from the beginning of the dialed string before the call is pl via the selected trunk. Example: The users will dial 9 as the first digit of long-distance calls. However, 9 should not be sent out via digital lines and the PSTN line this case, 1 digit should be stripped before the call is placed.
Prepend	Specify the digits to be prepended before the call is placed via the trunk. Those digits will be prepended after the dialing number stripped.
Time Condition	
Time Condition Mode	Use Main Trunk or Failover Trunk: Use the Main Trunk and its settings during the configured time conditions. If the main trunk i unavailable, the Failover Trunk and its settings will be used instead. Use Specific Trunks: Use specific trunks during the configured time conditions. The Strip and Prepend settings of the Main Trun
Time Condition	be used. If a trunk is unavailable during its time condition, no failover trunks will be used. Users could customize holiday time, office time, or a specified time to allow the outbound route to be used.

Inbound Routes

When a call comes into the GXW450X from the outside, it will usually arrive along with information about the telephone number that was dialed (a known as the "DID") and the Caller ID of the person who called.

The Inbound Routes are used to tell the system what to do with calls that come into the GXW450X on any trunk based on the pattern of the DID α caller ID of the person who called.

Inbound routes can be configured via Web GUI→ Trunks→Inbound Routes.



Figure 51: Create Inbound Routes

Inbound Route Configuration

Trunks	Select the trunk to configure the inbound rule.

	 All patterns are prefixed with the "_". Special characters: 				
Pattern	·				
	X: Any Digit from 0-9.				
	Z : Any Digit from 1-9.				
	N: Any Digit from 2-9.				
	".": Wildcard. Match one or more characters.				.,
	"!": Wildcard. Match zero or more characters immediately.				_X.
	Example: [12345-9] – Any digit from 1 to 9.		Pat ter	CallerID Pattern	_NXXXX
	Notes:		n		X
	 Multiple patterns can be used. Each pattern sh Example: 	ould be entered in new line.			_818X
	Pattern	CallerID Pattern			
	rattern	Callerid Pattern			
	_X.	1000			
	_NNXXNXXXX	1001			
	_818X.				
	Callant D. Data and				
Pattern	CallerID Pattern				
	CallerID Pattern				
_X.	CallerID Pattern				
_X. _NNXXN					
Pattern _X. _NNXXN XXXXX _818X.	1000				
_XNNXXN xxxxx _818X.	1000				
XNNXXN XXXXXX _818X. Pattern	1000				
X. NNXXN XXXXX 818X. Pattern X.	1000				
X. NNXXN (XXXX 818X. Pattern X. NNXXN (XXXX	1000 1001 CallerID Pattern				
_X. _NNXXN XXXXX	1000 1001 CallerID Pattern 1000 1001				
_XNNXXN xXXXXX _818X. Pattern _XNNXXN xXXXXX	1000 1001 CallerID Pattern 1000 1001	e do not enter more than one "_" at the beginning. In			
X. NNXXN XXXXX 818X. Pattern X. NNXXN XXXXX	1000 CallerID Pattern 1000 1001 All patterns are prefixed by "_" character, but pleas				
X. NNXXN XXXXXX 818X. Pattern X. NNXXN XXXXX	1000 CallerID Pattern 1000 1001 All patterns are prefixed by "_" character, but pleas patterns, some characters have special meanings:				
_XNNXXN xXXXXX _818X. Pattern _XNNXXN xXXXXX _818X.	1000 CallerID Pattern 1000 1001 All patterns are prefixed by "_" character, but pleas patterns, some characters have special meanings: [12345-9] Any digit in the brackets. In this exam				
_XNNXXN xXXXXX _818X. Pattern _XNNXXN xXXXXX	1000 CallerID Pattern 1000 1001 All patterns are prefixed by "_" character, but pleas patterns, some characters have special meanings: [12345-9] Any digit in the brackets. In this exam N Any digit from 2-9.	ole, 1,2,3,4,5,6,7,8,9 are allowed.			
_XNNXXN XXXXXX _818X. Pattern _XNNXXN XXXXXX _818X.	1000 CallerID Pattern 1000 1001 All patterns are prefixed by "_" character, but pleas patterns, some characters have special meanings: [12345-9] Any digit in the brackets. In this exam N Any digit from 2-9. Wildcard, matching one or more characters.	ole, 1,2,3,4,5,6,7,8,9 are allowed.			
X. ANNXXN B18X CallerID	1000 CallerID Pattern 1000 1001 All patterns are prefixed by "_" character, but pleas patterns, some characters have special meanings: [12345-9] Any digit in the brackets. In this exam N Any digit from 2-9. Wildcard, matching one or more characters. ! Wildcard, matching zero or more characters in	ole, 1,2,3,4,5,6,7,8,9 are allowed.			
_XNNXXN XXXXXX _818X. Pattern _XNNXXN XXXXXX _818X.	1000 CallerID Pattern 1000 1001 All patterns are prefixed by "_" character, but pleas patterns, some characters have special meanings: [12345-9] Any digit in the brackets. In this exam N Any digit from 2-9. Wildcard, matching one or more characters. ! Wildcard, matching zero or more characters im X Any digit from 0-9.	ole, 1,2,3,4,5,6,7,8,9 are allowed. mediately.			

Set Caller ID Info	Manipulates Caller ID (CID) name and/or number within the call flow to help identify who is calling. When enabled two field will show allowing to manipulate the CallelD Number and the Caller ID Name.
CalleID Number	Configures the pattern-matching format to manipulate the numbers of incoming callers or to set a fixed callerID number for calls that go through this inbound route. • \${CALLERID(num)}: Default value which indicates the number of an incoming caller (CID). The CID will not be modified. • \${CALLERID(num):n}: Skips the first n characters of a CID number, where n is a number. • \${CALLERID(num):-n}: Takes the last n characters of a CID number, where n is a number. • \${CALLERID(num):s:n}: Takes n characters of a CID number starting from s+1, where n is a number and s is a character position (e.g. \${CALLERID(num):2:7} takes 7 characters after the second character of a CID number). n\${CALLERID(num)}: Prepends n to a CID number, where n is a number.
CallerID Name	Default string is \${CALLERID(name)} which means the name of a incoming caller, it's a pattern-matching syntax format. A\${CALLERID(name)}B means Prepend a character 'A' and suffix a character 'B' to \${CALLERID(name)}. Not using pattern-matching syntax means setting fix name to incoming caller.

Table 35: Inbound Rule Configuration Parameters

Inbound Route: Import/Export Inbound Route

Users can import and export inbound routes to quickly set up inbound routing on a GXW450X or to back up an existing configuration. An exporte inbound route configuration can be directly imported without needing any manual modifications.



Figure 52: Import/Export Inbound Route

The imported file should be in CSV format and using UTF-8 encoding, the imported file should contain the below columns, and each column shou separated by a comma (It is recommended to use Notepad++ for the imported file creation):

- o Pattern: Always prefixed with _
- CallerID Pattern: Always prefixed with _

GATEWAY SETTINGS

This section describes internal options that haven't been mentioned in previous sections yet. The settings in this section can be applied globally to GXW450X, including general configurations, jitter buffer, RTP settings, and hardware config. The options can be accessed via Web GUI->Gateway Settings->General Settings.

SIP Settings

The GXW450X SIP global settings can be accessed via Web GUI→ Gateway Settings→SIP Settings.

General

On this page, users can define the Binding UDP Port for SIP protocol and Enable 486 to Failover Trunk.



Figure 53: SIP Settings – General

Bind UDP Port	Configure binding UDP port for SIP. The default setting is "5060".
Enable 486 to Failover Trunk	Reroutes failed outbound calls that receive a 486 response through the failover trunk to retry the call. If disabled, calls t receive a 486 response will be terminated.
	The default setting is "unchecked".

Table 36: SIP Settings – General

Misc

On this Web page, users can define the DNS mode used by the GXW450X and Outbound SIP Registrations.

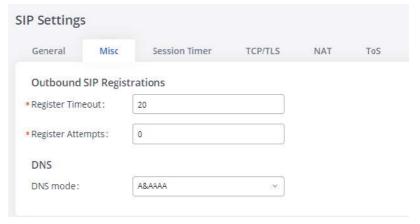
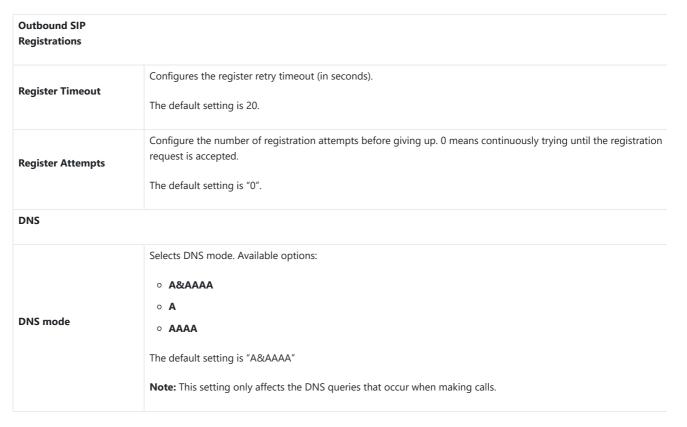


Figure 54: SIP Settings/Misc



Session Timer

Force Timer	If checked, always request and run the session timer. The default is "unchecked".
Timer	If checked, run the session timer only when requested by another UA. Default is "checked".
Session Expire	Configure the maximum session refresh interval (in seconds). The default setting is 1800.
Min SE	Configure the minimum session refresh interval (in seconds). The default setting is 90.

Table 38: SIP Settings/Session Timer

TCP and TLS

Configure to allow incoming TCP connections with the GXW450X. The default setting is "No".
Configure the IP address for the TCP server to bind to. 0.0.0.0 means binding to all interfaces. The port number is optional. If not spe 5060 will be used. The default setting is "0.0.0.0:5060"
Configure the IPv6 address for the TCP server to bind to. "[::]" means bind to all interfaces. The port number is optional with the defa being 5060. For example, [2001:0DB8:0000:0000:0000:1428:0000]:5060. The default setting is "[::]:5060".
Configure to allow incoming TLS connections with the GXW450X. The default setting is "No".
Configure the IP address for the TLS server to bind to. 0.0.0.0 means binding to all interfaces. The port number is optional. If not specified will be used. The default setting is "0.0.0.0:5061". Note: The IP address must match the common name (hostname) in the certificate. Please do not bind a TLS socket to multiple IP addresses. For details on how to construct a certificate for SIP, please refer to the following document: http://tools.ietf.org/html/draf sip-domain-certs
Configure the IPv6 address for the TLS server to bind to. "[::]" means bind to all interfaces. The port number is optional with the defa being 5061. For example, [2001:0DB8:0000:0000:0000:0000:1428:0000]:5061. Note: The IP address must match the common name (hostname) in the certificate so that the TLS socket won't bind to multiple IP addresses. The default setting is "[::]:5061".
If enabled, the TLS server's certificate won't be verified when acting as a client. The default setting is "Yes".
This is the CA certificate if the TLS server being connected to requires a self-signed certificate, including the server's public key. This I will be renamed as "TLS.ca" automatically. Note: The size of the uploaded "ca" file must be under 2MB.

TLS Cert	This is the Certificate file (*.pem format only) used for TLS connections. It contains a private key for the client and a signed certificate the server. This file will be renamed as "TLS.pem" automatically. Note: The size of the uploaded certificate file must be under 2MB.
TLS Key	The size of a private key must be under 2MB. This is the private key (*.key format only) for TLS connections. This file will be renamed as "TLS.key" automatically.
TLS CA Cert	This file must be named with the CA subject name hash value. It contains CA's (Certificate Authority) public key, which is used to veri accessed servers. Note: The size of the uploaded CA certificate file must be under 2MB.
TLS CA List	Display a list of files under the CA Cert directory.

Table 39: SIP Settings/TCP and TLS

The configuration in this section requires system reboot to take effect.

NAT

External Host	Configure a static IP address and port (optional) used in outbound SIP messages if the GXW450X is behind NAT. If it is a hostname, it will only be looked up once.
Use IP address in SDP	If enabled, the SDP connection will use the IP address resolved from the external host. The default setting is "enabled".
External UDP Port	Configure the externally mapped UDP port when the GXW450X is behind a static NAT or PAT. The default setting is "5060".
External TCP Port	Configure the externally mapped TCP port when the GXW450X is behind a static NAT or PAT. The default setting is "5060".
External TLS Port	Configures the externally mapped TLS port when GXW450X is behind a static NAT or PAT. The default setting is "5061".
Local Network Address	Adds a list of network addresses that are considered inside of the NAT network. Multiple entries are allowed. If not configur the external IP address will not be set correctly. A sample configuration could be as follows: 192.168.0.0/16

Table 40: NAT Settings

ToS

ToS For SIP	Configure the Type of Service for SIP packets. The default setting is None.
ToS For RTP Audio	Configure the Type of Service for RTP audio packets. The default setting is None.
Send Compact SIP Headers	If enabled, compact SIP headers will be sent. The default setting is "No". Note: This change requires a system reboot to take effect.
Enable Relaxed DTMF	Select to enable relaxed DTMF handling. The default setting is "No".

	Configures the default mode for sending DTMF.
	• RFC2833: DTMF is transmitted as audio in the RTP stream but is encoded separately from the audio stream.
	o Inband: DTMF is transmitted as audio and is included in the audio stream. Requires alaw/ulaw codecs.
DTMF Mode	SIP INFO: DTMF is transmitted through a separate network connection from the media streams.
	Auto: DTMF mode will be negotiated with the remote peer. RFC2833 will be used by default unless the remote.
	peer does not support it.
	The default setting is "RFC2833".
RTP Timeout	Configure the timeout in seconds. When the call is in talking status, if there is no RTP activity after the timeout, the
	will be terminated. The default setting is 90 seconds.
RTP Hold Timeout	Configure the timeout in seconds. When the call is on hold, if there is no RTP activity after the timeout, the call wil
	terminated. This value must be larger than "RTP Timeout". The default setting is 100 seconds.
RTP Keep-alive	Configure the interval (in seconds) that an RTP Keepalive packet will be sent on an SDP connection. The default se
KIF Keep-alive	is 0 (no RTP Keepalive).
Default	
Incoming/Outgoing	Configure the default duration (in seconds) of incoming/outgoing registration. The default setting is 120.
Registration Time	
	Configure the 100rel setting on GXW450X.
	No: Unsupported.
100rel	• Yes: Supported.
100101	Required: Forced to support.
	The default setting is "Yes".
Trust Remote Party ID	Configure whether the Remote-Party ID should be trusted. The default setting is "No".
Cond Domesto Doube ID	Confirmation the Devices Device ID should be expected at the default and in its White
Send Remote Party ID	Configure whether the Remote-Party ID should be sent or not. The default setting is "No".
	Configure whether the GXW450X should generate inband ringing or not.
	Yes: The GXW450X will send 180 Ringing followed by 183 Session Progress and in-band audio.
	• No: The GXW450X will send 180 Ringing if 183 Session Progress has not been sent yet. If the audio path is
Generate In-Band Ringing	established already with 183 then send in-band ringing.
	• Never: Whenever ringing occurs, the GXW450X will send 180 Ringing as long as 200OK has not been set yet.
	The default setting is "Never".
Server User Agent	Configure the user agent string for the GXW450X.
Default	
Incoming/Outgoing	Configure the default duration (in seconds) of incoming and outgoing registration. The default setting is 120 seco
Registration Time	

Table 41: ToS Settings

RTP Settings

RTP Settings

RTP Start	Configure the RTP port starting number. The default setting is 10000.	
RTP End Configure the RTP port ending address. The default setting is 20000.		

Strict RTP	Configure to enable or disable strict RTP protection. If enabled, RTP packets that do not come from the source of the RTP stream be dropped. The default setting is "Disable".
RTP Checksums	Configure to enable or disable RTP Checksums on RTP traffic. The default setting is "Disable".

Table 42: RTP Settings

Payload Type Settings

The GXW450X payload type for audio codecs can be configured here.

AAL2-G.726	Configure payload type for ADPCM (G.726, 32kbps, AAL2 codeword packing). The valid range is between 96 and 127. The de setting is 112.
DTMF	Configure payload type for DTMF. The valid range is between 96 and 127. The default setting is 101.
G.721 Compatible	Configure to enable/disable G.721 compatible. The default setting is Yes.
G.726	Configure the payload type for G.726 if "G.721 Compatible" is disabled. The default setting is 111.
iLBC	Configure the payload type for iLBC. The valid range is between 96 and 127. The default setting is 97.
OPUS	Configure the payload type for OPUS. The valid range is between 96 and 127. The default setting is 123.

Table 43: Payload Type Configuration

- Click on Default All to set the values of the payload parameters to the factory default values
- While configuring the payload values users can Click on Reset All to reset the values to the last saved values on the gateway.

Voice Prompt

The GXW450X supports multiple languages in Web GUI as well as system voice prompt. The following languages are currently supported in the sy voice prompt:

English (United States), British English, Arabic, Chinese, Dutch, French, German, Greek, Hebrew, Italian, Polish, Portuguese, Russian, Spanish, Catala Swedish, Czech, and Turkish.

English (United States) and Chinese voice prompts are built-in with the GXW450X already. The other languages provided by Grandstream can be downloaded and installed from the GXW450X Web GUI directly. Additionally, users could customize their own voice prompts, package them and ι them to the GXW450X.

Language settings for voice prompts can be accessed under Web GUI → Gateway Settings → Voice Prompt → Language.

Download and Install Voice Prompt Package

To download and install voice prompt package in different languages from GXW450X Web GUI, click on Check Prompt List button.

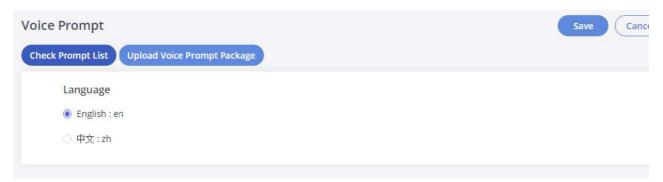


Figure 55: Language Settings for Voice Prompt

A new dialog window of the voice prompt package list will be displayed. Users can see the version number (latest version available V.S. current ins version), package size, and options to upgrade or download the language

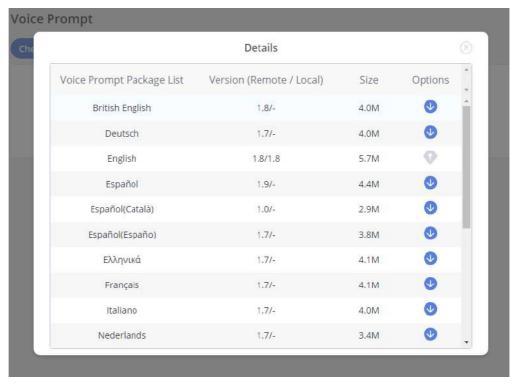


Figure 56: Voice Prompt Package List

Click on • to download the language to the GXW450X. The installation will be automatically started once the downloading is finished.

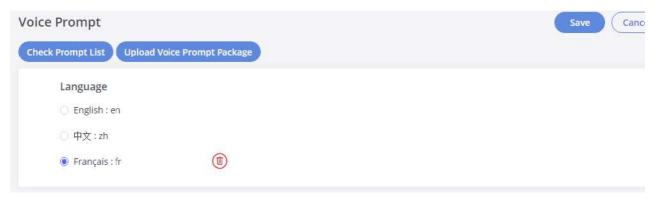


Figure 57: New Voice Prompt Language Added

A new language option will be displayed after successfully installed. Users then could select it to apply in the GXW450X system voice prompt or d from the GXW450X

Manual Upload of Prompt Package

Users can upload the prompt package manually to the GXW450X. Users can create their own prompt package for different languages and use the the default voice prompts.

To upload the voice prompt to the GXW450X, press the Upload Voice Prompt Package button and brows the prompt package.

Voice Prompt
Check Prompt List Upload Voice Prompt Package

Language

English: en

中文: zh

Figure 58: Upload Voice prompt Package

Call Failure Tone Settings

SIP Trunk Prompt Tone

Prompt Tone Settings tab has been added to the GXW to help users choose which prompt will be played by the GXW during call failure, the follow voice message responses have been added and can be set to be played for 4XX, 5XX, and 6XX call failures:

- o The default for 404 and 604 status codes: "Your call can't be completed as dialed. Please check the number and dial again."
- The default for 5xx status codes: "Server error. Please check your device."
- o The default for 403 and 603 status codes: "The call was rejected by the server. Please try again later."
- o The default for all other status codes: "All circuits are busy now. Please try again later."

Additionally, custom voice messages recorded and uploaded in Gateway Settings >Voice Prompt >Custom Prompt can be used for these failure responses instead of the default messages.

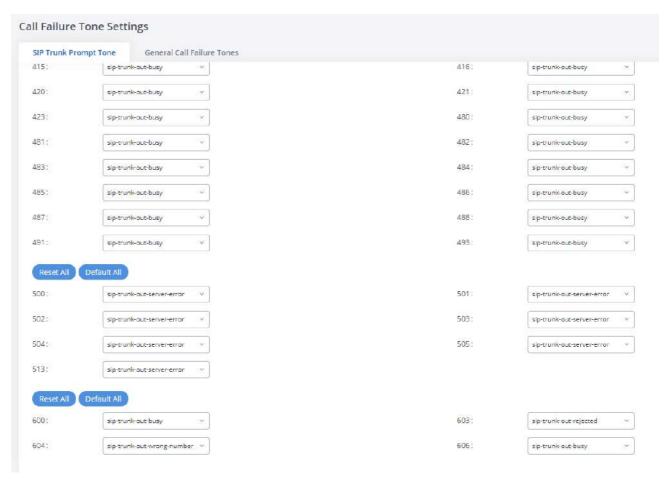


Figure 59: SIP Trunk Prompt Tone

General Call Failure Tone

Moreover, users also have the possibility to customize the prompt for typical call failure reasons like (no permission to allow outbound calls, busy incorrect number dialed ...etc.).

To customize these prompts users could record and upload their own files under "Gateway Settings → Voice Prompt → Custom Prompts" then sel each one for a specific call failure case under the "Gateway Settings → Call Failure Tone Settings → General Call Failure Tones" page as shown on the following figure:

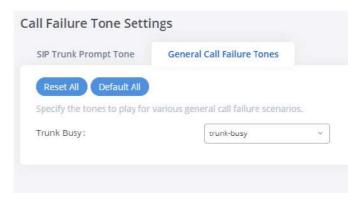


Figure 60: General call Failure Tones

Jitter Buffer

A jitter buffer is used at the receiving equipment to store incoming RTP packets, re-align them in terms of timing and check they are in the correct If some arrive slightly out of sequence then, provided it is large enough, the jitter buffer can put them back into the right sequence. However, for twork the receiving device must delay the audio very slightly while it checks and reassembles the packet stream.

Below are the Jitter buffer Settings to control the size of the buffer and its implementation mode:

Enable Jitter Buffer	Select to enable the jitter buffer on the sending side of the SIP channel. The default setting is "No".	
Jitter Buffer Size	Configure the time (in ms) to buffer. This is the jitter buffer size used in the "Fixed" jitter buffer or used as the initial time for the "adaptive" jitter buffer. The default setting is 100.	
Implementati on	Configure the jitter buffer implementation on the sending side of a SIP channel. The default setting is "Fixed". • Fixed The size is always equal to the value of "Max Jitter Buffer". • Adaptive The size is adjusted automatically and the maximum value equals the value of "Max Jitter Buffer".	
Max Jitter Buffer	Configure the maximum time (in ms) to buffer for "Adaptive" jitter buffer implementation or used it as the jitter buffer size for " jitter buffer implementation. The default setting is 100.	

Table 44: Jitter Buffer Settings

Interface Settings

The GXW450X supports E1/T1/J1 which are physical connection technologies used in digital networks. T1 is the North American standard, J1 is used Japan, whereas E1 is the European standard. GXW450X supports four signaling protocols: PRI_NET, PRI_CPE, MFC/R2, and SS7. PRI provides a vary number of channels depending on the standards in the country of implementation (E1, T1, or J1); MFC/R2 is a signaling protocol heavily used ove trunks; SS7 uses out-of-band signaling, which travels on a separate, dedicated channel rather than within the same channel as the telephone call, providing more efficiency and higher security level when the telephone calls are set up.

The interface settings page allows the configuration of digital hardware parameters. For more details, refer to the **[Digital Hardware Configurati** section.

GXW450X also allows the users to configure **Tone Region** by choosing their country to set the default tones for dial tone, busy tone, and ring ton be sent. If not specified, the default setting is "The United States".

MAINTENANCE

The Maintenance section lists different tools to help troubleshoot the issues that might be encountered while using the GXW450X alongside a set options to manage users, control web GUI access, upgrade the firmware, backup the configuration, take ethernet and Digital traces ...etc.

User Management

User management is on the Web GUI→ **Maintenance→User Management** page. Users could create multiple accounts for different administrator in to the GXW450X Web GUI.

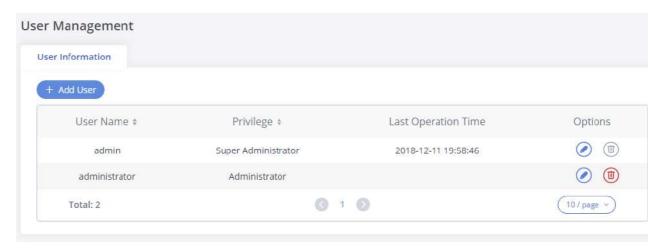


Figure 61: User Management Page Display

- When logged in as Super Admin, click on + Add User to create a new account for Web GUI user. The following dialog will prompt. Configure 1 parameters as shown in below table.

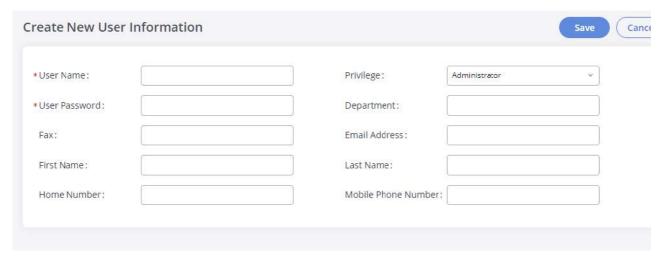


Figure 62: Create New User

User Name	Configures a username to identify the user which will be required in Web GUI login. Letters, digits, and underscores are allowed in the user name.	
Privilege	This is the role of the Web GUI user. Currently, only "Admin" is supported when Super Admin creates a new user.	
User Password	Configures a password for this user which will be required in Web GUI login. Letters, digits, and underscores are allowed.	
Department	Enters the necessary information to keep a record for this user.	
Fax		
Email Address		
First Name		
Last Name		

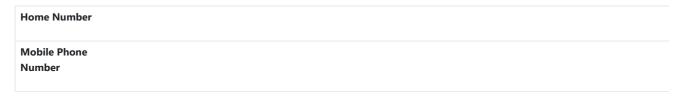


Table 45: Create New User Information

Change Information

Change Password

Follow the steps below to change the Web GUI access password.

- 1. Go to the Web GUI→Maintenance→Change Information page.
- 2. Enter the old password first.
- 3. Enter the new password and retype the new password to confirm. The new password has to be at least 4 characters. The maximum length of t password is 16 characters.
- 4. Configure the Email Address that is used when login credentials are lost.
- 5. Click on "Save" and the user will be automatically logged out.
- 6. Once the web page comes back to the login page again, enter the username "admin" and the new password to login.

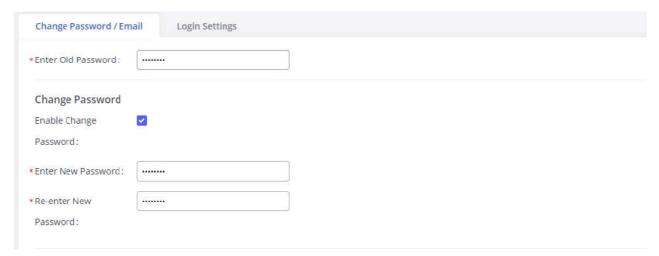


Figure 63: Change Password

Enter Old Password	Enter the old Password for GXW450X
Enable Change Password	When enabled, the fields to enter the new password will be displayed
Enter New Password	Enter the New Password for GXW450X
Re-enter New Password	Retype the New Password for GXW450X

Table 46: Change Password Parameters

Change Binding Email

GXW450X allows users to configure binding email in case log in password is lost. GXW450X login credentials will be sent to the designated email address. The feature can be found under Web GUI->Maintenance->Change Information->Change Binding Email.

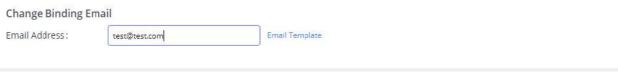


Figure 64: Change Binding Email

Login Settings

After the user logs in to the GXW450X Web GUI, the user will be automatically logged out after a certain timeout, or he/she can be banned for a s period if the login timeout is exceeded. Those values can be specified under the GXW450X web GUI > Maintenance > Change Information > Log Settings page.

The "User Login Timeout" value is in minutes and the default setting is 10 minutes. If the user doesn't make any operation on Web GUI within the timeout, the user will be logged out automatically. After that, the Web GUI will be redirected to the login page and the user will need to enter the username and password to log in.

If set to 0, there is no timeout for the Web GUI login session and the user will not be automatically logged out.

The "maximum number of login attempts" can prevent the GXW450X from brute force decryption, if this number is exceeded user IP address w banned from accessing the GXW for a period based on user configuration, the default value is 5.

"User ban period" specifies the period in minutes an IP will be banned from accessing the GXW if the User max number of try login is exceeded, to default value is 5.

"Login Banned User List" shows the list of IPs banned from the GXW.

"Login White List" Users can add a list of IPs' to avoid the above restriction, thus, they can exceed the User's max number of try login.

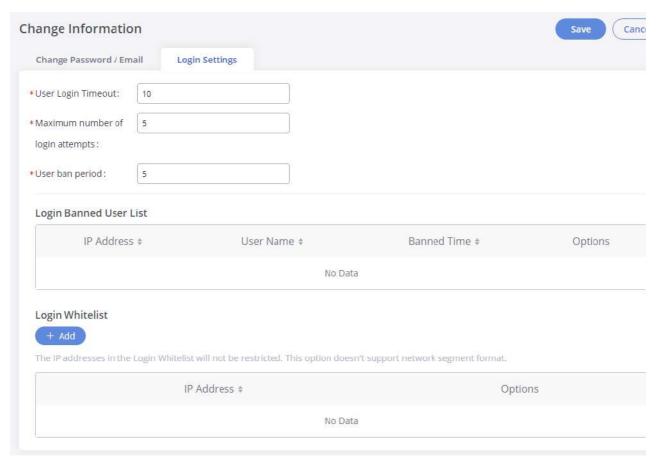


Figure 65: Login Timeout Settings

Operation Log

The admin has the authority to view operation logs on the GXW450X Web GUI >Maintenance > Operation Log page. Operation logs list the operation by all the Web GUI users, for example, Web GUI login, creating trunk, creating outbound rule, etc. There are 7 columns to record the operation details "Date", "User Name", "IP Address", "Results", "Page Operation", "Specific Operation" and "Remark".

Operation Log m Delete Search Result (s) Delete All Logs Filter User Na IP Addres Date # Results # Page Operation # Specific Operation \$ Remark # me # S # 2018-12-12 0 Operation su admin 172.16.1.62 downloadFile type: syslog. (i) 2:17:02 ccessful 2018-12-12 0 Operation su 172,16,1,62 User Name: admin. ① admin Login: Login 2:16:55 ccessful Wrong accou 2018-12-12 0 User Name: admin. ① admin 172.16.1.62 nt or passwo Login: Login 2:16:48 Wrong accou 2018-12-12 0 User Name: admin. ① admin 172.16.1.62 nt or passwo Login: Login 2:16:46 rd! Wrong accou 2018-12-12 0 User Name: admin. ① admin 172.16.1.62 nt or passwo Login: Login 2:16:41 rd! 2018-12-11 2 41.250.164.9 Operation su User Name: admin. ① admin Login: Login 3:06:34 4 ccessful 2018-12-11 2 41.250.164.9 Operation su User Name: admin. ① admin Login: Login 2:51:17 ccessful 2018-12-11 2 41.250.164.9 Operation su admin Logout: Logout User Name: undefined. ① 2:50:59 4 ccessful 2018-12-11 2 41.250.164.9 Operation su admin Login: Login User Name: admin. ① 2:41:13 4 ccessful 2018-12-11 2 41.250.164.9 Operation su Extensions: Update User Inf admin User Name: admin. ① 2:40:57 ccessful ormation Total: 181 (3) 1 2 3 4 5 ··· 19 () 10 / page v

Figure 66: Operation Logs

The operation log can be sorted and filtered for easy access. Click on the header of each column to sort. For example, clicking on "Date" will sort t according to the operation date and time. Clicking on "Date" again will reverse the order.

Date	The date and time when the operation is executed.
User Name	The username of the user who performed the operation.
IP Address	The IP address from which the operation is made.
Results	The result of the operation.
Page Operation	The page where the operation is made. For example, login, logout, delete user, create trunk and etc.
Specific Operation	Click on ® to view the options and values configured by this operation.
Remark	Allows users to add notes and remarks to each operation

Users could also filter the operation logs by time condition, IP address and/or username. To use the filter, click on conditions then click on Search .

Filter ^ and configure the

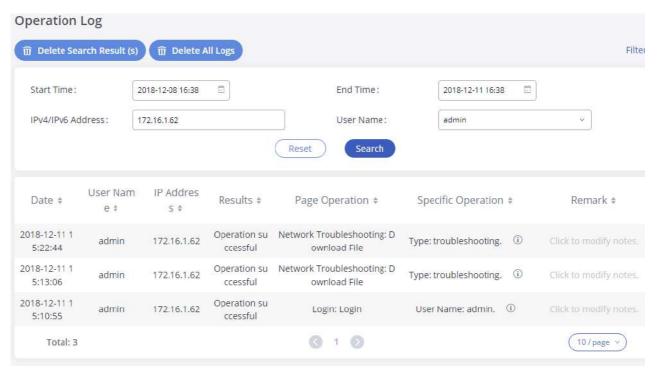


Figure 67: Operation Logs Filter

The above figure shows an example that operations made by user "admin" on a device with IP 172.16.1.62 from 2018-12-08 16:38 to 2018-12-11 are filtered out and displayed.

To delete operation logs, users can perform filtering first and then click on Delete Search Result (s) to delete the filtered result of operation Or users can click on Delete All Logs to delete all operation logs at once.

Syslog

On the GXW450X, users could dump the Syslog information to a remote server under Web GUI **> Maintenance -> Syslog**. Enter the Syslog server hostname or IP address and select the module/level for the Syslog information.

The default Syslog level for all modules is "error", which is recommended in your GXW450X settings because it can be helpful to locate the issues errors happen.

Some typical modules for GXW450X functions are as follows and users can turn on "notice" and "verb" levels besides the "error" level.

- $\circ~$ $\mbox{\bf pbx:}$ This module is related to general PBX functions.
- o **pjsip:** This module is related to SIP calls.
- $\circ \ \ \textbf{chan_dahdi:} \ \text{This module is related to digital calls (E1/T1/J1)}.$

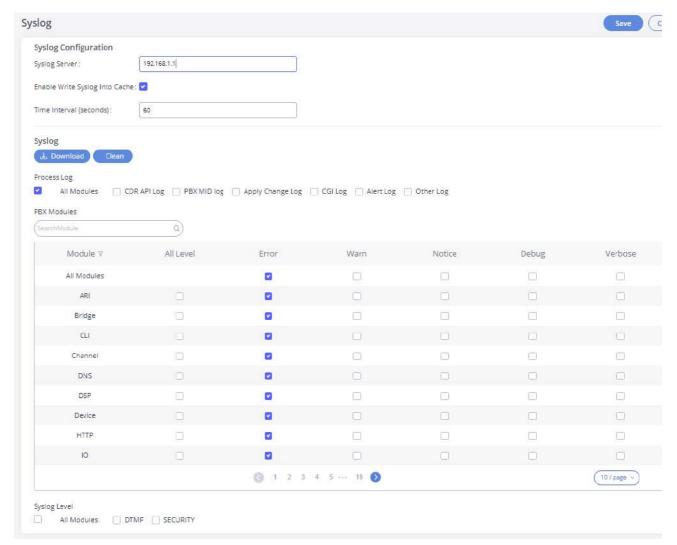


Figure 68: Syslog Settings

Syslog is usually for debugging and troubleshooting purpose. Turning on all levels for all Syslog modules is not recommended for daily usage. Too many Syslog prints might cause traffic and affect system performance.

The reserved size for Syslog entries on the cache memory of the GXW is 50M, once this size is reached the GXW will clean up 2M of the oldest Sys entries to allow to save new logs.

System Events

The GXW450X can monitor important system events, log the alerts and send Email notifications to the system administrator.

Alert Log

Under Web GUI → Maintenance → System Events → Alert Log, system messages are listed when the alert is triggered for the configured system ev The following picture shows the "User Login Successes", "User Login Failed" and "System Reboot" alert logs.

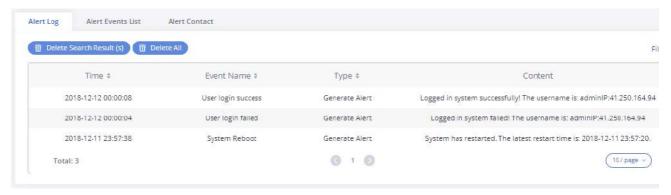


Figure 69: System Events →Alert Log

Users could also filter the Alert Logs by time condition, Event Name, and/or Type. To use the filter, click on then click on search.

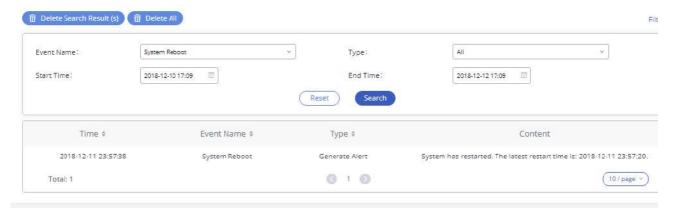


Figure 70: Alert Log Filter

The above figure shows an example of a System reboot Alerts logged on 2018-12-11 at 23:57 displayed using the filter Event name System Reboc

To delete alert logs, users can perform filtering first and then click on users can click

Alert Events List

The system alert events list can be found under Web GUI **Maintenance System Events**. The following event is currently support the GXW450X which will have an alert, and/or an Email generated if occurred:

- o Disk Usage
- o Modify Super Admin Password
- Memory Usage
- o System Reboot
- System Update
- System Crash
- o Register SIP trunk failed
- Restore Config
- User Login Success
- User Login Failed
- o SIP Outgoing Call through Trunk Failure
- o Fail2ban Blocking
- o SIP Peer Trunk Status
- User Login Banned
- o External Disk Usage
- o The CDR database is corrupted

Click on o to configure the parameters for each event

1. Disk Usage



Figure 71: System Events →Alert Events Lists: Disk Usage

- Detect Cycle: The GXW450X will perform the internal disk usage detection based on this cycle. Users can enter the number and then select second(s)/minute(s)/hour(s)/day(s) to configure the cycle.
- o Alert Threshold: If the detected value exceeds the threshold (in percentage), the GXW450X system will send the alert.

Alert Settings: External Disk Usage



Figure 72: System Events →Alert Events Lists: External Disk Usage

- **Detect Cycle**: The GXW450X will perform the External disk usage detection based on this cycle. Users can enter the number and then select second(s)/minute(s)/hour(s)/day(s) to configure the cycle.
- o Alert Threshold: If the detected value exceeds the threshold (in percentage), the GXW450X system will send the alert.
- 3. Memory Usage



Figure 73: System Events ->Alert Events Lists: Memory Usage

- **Detect Cycle**: The GXW450X will perform the memory usage detection based on this cycle. Users can enter the number and then select second(s)/minute(s)/hour(s)/day(s) to configure the cycle.
- o Alert Threshold: If the detected value exceeds the threshold (in percentage), the GXW450X system will send the alert.
- 4. System Crash



Figure 74: System Events -> Alert Events Lists: System Crash

• **Detect Cycle**: The GXW450X will detect the event at each cycle based on the specified time. Users can enter the number and then select second(s)/minute(s)/hour(s)/day(s) to configure the cycle.

Click on the switch or to turn on/off the alert and Email notification for the event. Users could also select the checkbox for each eventhen click on buttons "Alert On", "Alert Off", "Email Notification On", or "Email Notification Off" to control the alert and Email notification configuration of the alert and Email notification configuration of the alert and Email notification of the alert and Email notification of the event.

Alert Contact

Users could add the administrator's Email address under Web GUI→Maintenance→System Events→Alert Contact to send the alert notification. Up Email addresses can be added.

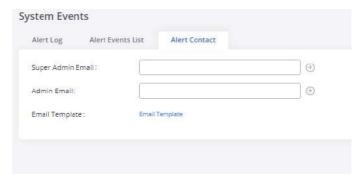


Figure 75: Alert Contact

Upgrade

The GXW450X can be upgraded to a new firmware version remotely or locally. This section describes how to upgrade your GXW450X via network upload.

Upgrading via Network

The GXW450X can be upgraded via TFTP/HTTP/B by configuring the URL/IP Address for the TFTP/HTTP/HTTPS server and selecting a downlow method. Configure a valid URL for TFTP, HTTP, or HTTPS; the server name can be FQDN or IP address.

The upgrading configuration can be accessed via Web GUI → Maintenance → Upgrade.



Figure 76: Network Upgrade

Upgrade Via	Allow users to choose the firmware upgrade method: TFTP, HTTP, or HTTPS.
Firmware Server Path	Configures firmware server path. For example, firmware.grandstream.com
Firmware File Prefix	If configured, only the firmware with the matching encrypted prefix will be downloaded.
Firmware File Suffix	If configured, only the firmware with the matching encrypted postfix will be downloaded.
HTTP/HTTPS User Name	The user name for the HTTP/HTTPS server.
HTTP/HTTPS Password	The password for the HTTP/HTTPS server.

Table 48: Network Upgrade Configuration

Please follow the steps below to upgrade the firmware remotely.

- 1. Enter the firmware server path under Web GUI→Maintenance→Upgrade.
- 2. Click on "Save". Then reboot the device to start the upgrading process.
- 3. Please be patient during the upgrading process. Once done, a reboot message will be displayed in the LCD.
- 4. Manually reboot the GXW450X when it's appropriate to avoid immediate service interruption. After it boots up, log in to the Web GUI to chec firmware version.

Upgrading via Local Upload

If there is no HTTP/TFTP server, users could also upload the firmware to the GXW450X directly via Web GUI. Please follow the steps below to uplo firmware locally.

- 1. Download the latest GXW450X firmware file from the following link and save it on your PC: https://www.grandstream.com/support/firmware
- 2. Log in to the Web GUI as an administrator on the PC.
- 3. Go to Web GUI→Maintenance→Upgrade, upload the firmware file by clicking on from your PC. The default firmware file name is *qxw4500fw.bin* and select the firmware file name is *qxw4500fw.bin*



Figure 77: Upgrading Firmware Files

- 4. Wait until the upgrading process is successful and a window will be popped up in the Web GUI requesting to confirm the reboot of the GXW4 the changes to take effect.
- 5. Click on "OK" to reboot the GXW450X and check the firmware version after it boots up.
 - Please do not interrupt or power cycle the GXW450X during the upgrading process.
 - o The firmware file name allows the use of the special characters besides the following restricted characters: #\$ ^ & * + ()[]/;'|, < >?

Upgrading via a Local Server

Users can download a free TFTP, FTP, or HTTP server and conduct a local firmware upgrade. A free window version TFTP server is available for dow from:

http://www.solarwinds.com/products/freetools/free_tftp_server.aspx

http://tftpd32.jounin.net

Please check our website at https://www.grandstream.com/support/firmware for the latest firmware.

Instructions for local firmware upgrade via TFTP:

- 1. Unzip the firmware files and put all of them in the root directory of the TFTP server,
- 2. Connect the PC running the TFTP server and the GXW450X to the same LAN segment;
- 3. Launch the TFTP server and go to the File menu → Configure → Security to change the TFTP server's default setting from "Receive Only" to "Tran Only" for the firmware upgrade;
- 4. Start the TFTP server and configure the TFTP server in the GXW450X web configuration interface;
- 5. Configure the Firmware Server Path to the IP address of the PC;
- 6. Update the changes and reboot the GXW450X.

End users can also choose to download a free HTTP server from http://httpd.apache.org/ or use

Microsoft IIS web server.

No Local Firmware Server

For users that would like to use remote upgrading without a local TFTP/FTP/HTTP server, Grandstream offers a NAT-friendly HTTP server. This enables users to download the latest software upgrades for the gateway via this server. Please refer to the following webpage for the firmware server path to use: https://www.grandstream.com/support/firmware

Backup

The GXW450X configuration can be backed up locally or via the network. The backup file will be used to restore the configuration on GXW450X w necessary.

Backup/Restore

Users could back up the GXW450X configurations for restore purposes under Web GUI→Maintenance→Backup→Backup/Restore. Click on Add Backup to create a new backup. Then the following dialog will show:

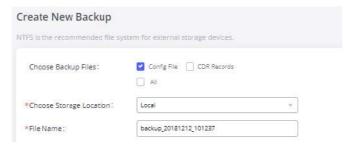


Figure 78: Create New Backup

- 1. Choose the files to be included in the backup.
- 2. Choose where to store the backup file: USB Disk, SD Card, or Local.
- 3. Name the backup file.
- 4. Click on "Backup" to start the backup.

Once the backup is done, the list of the backups will be displayed with the date and time on the web page. Users can download 👲, restore 👩, delete 🄞 it from the GXW450X internal storage or the external device.

Click on Upload Backup File to upload backup file from the local device to GXW450X. The uploaded backup file will also be displayed in the we and can be used to restore the GXW450X.

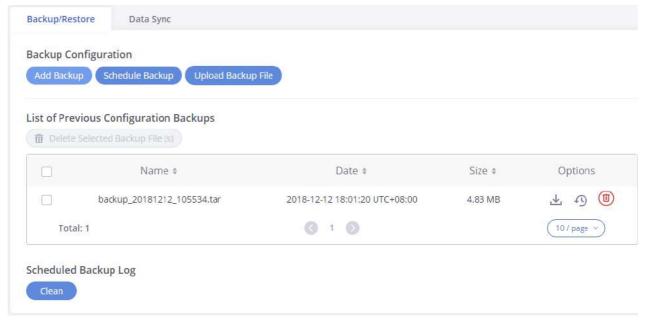


Figure 79: Backup / Restore

The Schedule Backup option allows GXW450X to perform automatic backup at the user-specified time. Scheduled backup files can only be stora uSB / SD card / SFTP server. Users can set backup time from 0-23 and how frequently the backup will be performed.

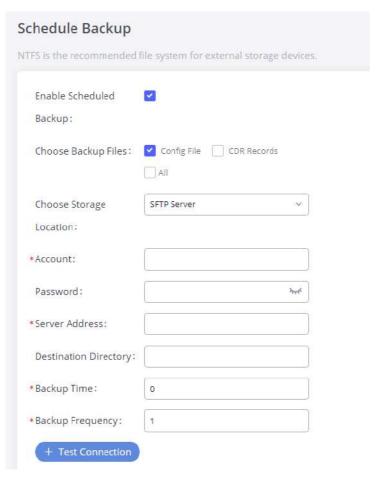


Figure 80: Schedule Backup

Data Sync

Besides local backup, users could backup the voice records and/or CDR on a daily basis to a remote server via SFTP protocol automatically under \GUI->Maintenance->Backup->Data Sync.

The client account supports special characters such as @ or ".". This change allows users to use an email address as SFTP accounts. It allows users to specify the destination directory on the SFTP server for the backup files. If the directory doesn't exist on the destination, GXW450X will create the directory automatically.

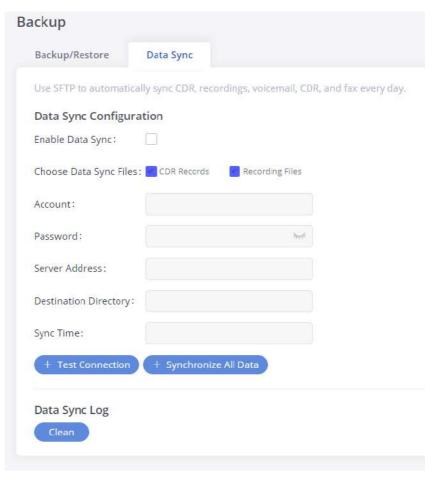


Figure 81: Data Sync

Enable Data Sync	Enable the auto backup function. This option is disabled by default				
Choose Data Sync Files	Choose the files to sync				
Account	Enter the Account name on the SFTP backup server.				
Password	Enter the Password associate with the Account on the SFTP backup server.				
Server Address	Enter the SFTP server address.				
Destination Directory	Specify the directory in the SFTP server to keep the backup file. Format: 'xxx/xxx/xxx', If this directory does not exist, GXW create this directory automatically.				
Sync Time	Enter 0-23 to specify the backup hour of the day.				

Table 49: Data Sync Configuration

Before saving the configuration, users could click on "Test Connection". The GXW450X will then try connecting the server to make sure the server and accessible for the GXW450X.

Save the changes and all the backup logs will be listed on the web page.

Restore Configuration from Backup File

To restore the configuration on the GXW450X from a backup file, users could go to Web GUI → Maintenance → Backup → Backup/Restore.

- A list of previous configuration backups is displayed on the web page. Users could click on 50 of the desired backup file and it will be restore the GXW450X.
- o If users have other backup files on the PC to restore on the GXW450X, click on "Upload Backup File" first and select it from the local PC to upl the GXW450X. Once the uploading is done, this backup file will be displayed in the list of previous configuration backups for restoring purpos Click on 49 to restore from the backup file.
- Users could also restore using the backup file saved on an SD card or USB device plugged into the GXW450X.

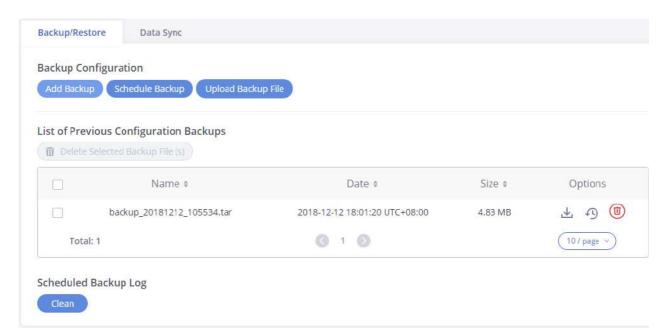


Figure 82: Restore GXW450X from Backup File

Backup file must be in tar format and contain letters, digits or special characters -_. The file size must be less than 10MB.

System Cleanup/Reset

Reset & Reboot

Users could perform reset and reboot under Web GUI → Maintenance → System Cleanup/Reset → Reset & Reboot. To factory reset the device, sel mode type first. There are two different types of reset.

- o User Data: The data such as CDR Records Operation Logs Core file etc.
- o All: Restore the device to factory default settings for both User Data and User Configuration.

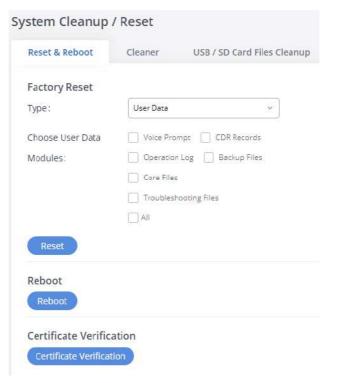


Figure 83: Reset and Reboot

- Press Reset to factory reset the GXW450X. Press Reboot to reboot the unit.
- Press Certificate Verification to validate certificate chain for the server's certificate.

Cleaner

Users could configure to clean the Call Detail Report/Voice Records/Voice Mails/FAX automatically under Web GUI→ Maintenance→System Cleanup/Reset →Cleaner.

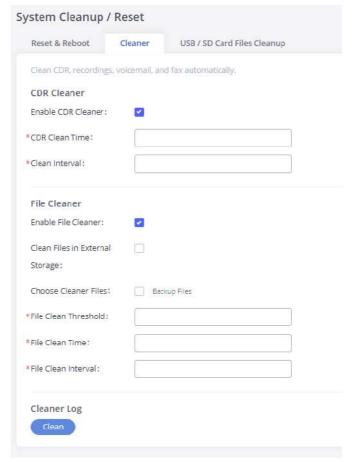


Figure 84: Cleaner

CDR Cleaner					
Enable CDR Cleaner	Enable the CDR Cleaner function.				
CDR Clean Time	Enter 0-23 to specify the hour of the day to clean up CDR.				
Clean Interval	Enter 1-30 to specify the day of the month to clean up CDR.				
File Cleaner					
Enable File Cleaner	Enter the Voice Records Cleaner function.				
Clean Files in External Device	If enabled the files in an external device (USB/SD card) will be atomically cleaned up as configured.				
Choose Cleaner File	Select the files for system automatic clean.				
File Clean Threshold	Specify the threshold of local storage usage from 0 to 99 (in percentage).				
File Clean Time	Enter 0-23 to specify the hour of the day to clean up the files.				
File Clean Interval	Enter 1-30 to specify the day of the month to clean up the files.				
Cleaner Log	Press the "Clean" button to clean the cleaner log.				

USB/SD Card Files Cleanup

Users could manage the content of the external drives, USB and /or SD card, manually from the Web GUI under **Maintenance→System Cleanup**/
→USB / SD Card Files Cleanup.

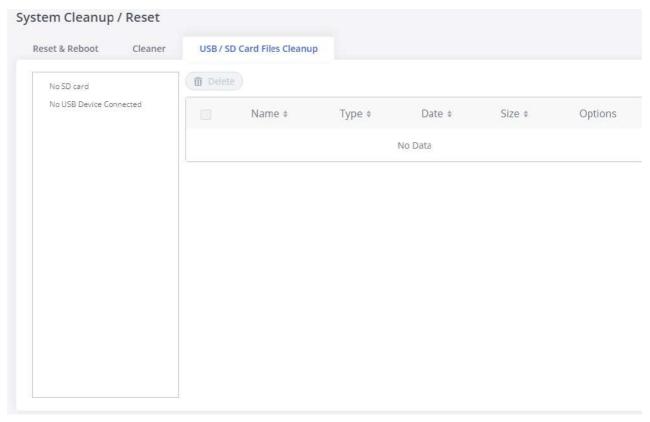


Figure 85: SB/SD Card Files Cleanup

On this Web page, users could navigate through the paths and the directories of the USB and/or the SD card and select the files and folders to cle

Network Troubleshooting

On the GXW450X, users could capture traces, ping remote host and traceroute remote host for troubleshooting purposes under Web GUI->Maintenance->Network Troubleshooting.

Ethernet Capture

An ethernet trace can be captured for troubleshooting purposes related to network issues, the SIP flow, etc.

The captured trace can be downloaded for analysis. Instructions or results will be displayed in the Web GUI output result.

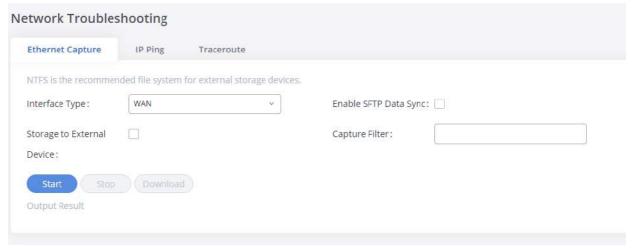


Figure 86: Ethernet Capture

Download	Click to download the trace if the trace is stored locally.
Stop	Click to stop the trace.
Start	Click to start the trace.
Capture Filter	Enter the filter to obtain the specific types of traffic, such as (host, src, dst, net, proto).
Storage to External Device	Check this box to activate storage of the capture either on the USB or SD Card.
Enable SFTP Data Sync	Check this box to save the capture files in the SFTP server. Please make sure the configuration of data synchronizatio works in advance.
Interface Type	Select the network interface to monitor.

Table 51: Ethernet Capture Parameters

The output result is in .pcap format. Therefore, users could specify the capture filter as used in the general network traffic capture tool (host, src, d protocol, port, port range) before starting to capture the trace.

IP Ping

Enter the Target Host using either a hostname or an IP address. Then press the "Start" button. The output result will dynamically be displayed in the window below.

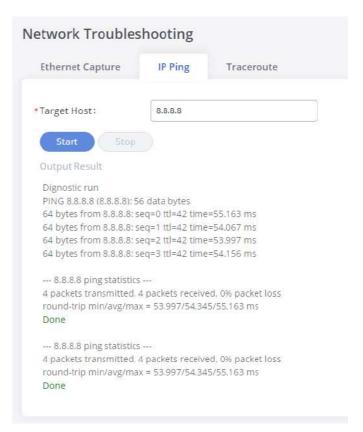


Figure 87: IP Ping

Traceroute

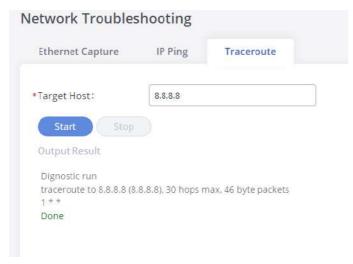


Figure 88: Traceroute

Signaling Troubleshooting

Please refer to the [Digital Trunk Troubleshooting] section.

Service Check

Enable Service Check to periodically check the GXW450X responsiveness. Check Cycle is configurable in seconds and the default setting is 60 sec. Times is the maximum number of failed checks before restarting the GXW450X. The default setting is 3. If there is no response from GXW450X aft attempts (default) to check, the current status will be stored and GXW450X will be restarted.

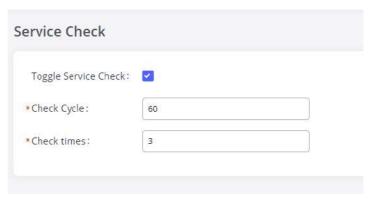


Figure 89: Service Check

CDR (CALL DETAIL RECORD)

CDR (Call Detail Record) is a data record generated by the PBX that contains attributes specific to a single instance of a phone call handled by the has several data fields to provide a detailed description of the call, such as the phone number of the calling party, phone number of the receiving start time, call duration, etc.

CDR Filter

On the GXW450X, the CDR can be accessed under Web GUI **CDR** Users could filter the call report by clicking on date range and criteria, depending on how the users would like to include the logs in the report. Click on the "Search" button to display the gener report.

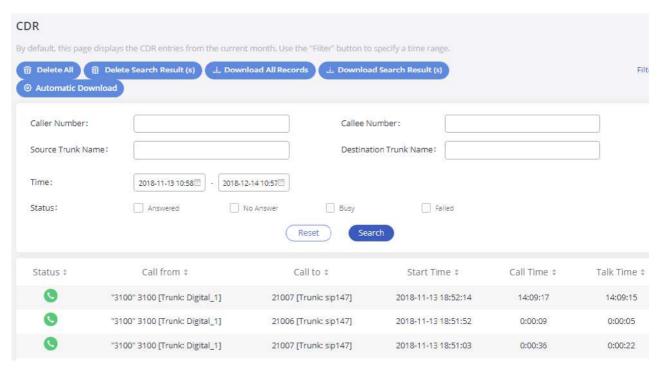


Figure 90: CDR Filter

Caller Number	You can specify a caller number or set a caller number with a pattern (. match zero or more characters only appears in the end. X any digit from 0-9, case-insensitive, repeatable, and only appears in the end. If the pattern string contains "." in the end, "X" must appear before ".".). For Example: X: It will filter out CDR records where a caller number is of ranges from 0 to 9. XXXX: It will filter out CDR records where a caller number has 4 digits. 3XXX: It will filter out CDR records where a caller number has a leading digit of 3 and a length of 4 digits. 3.: It will filter out CDR records where a caller number has a leading digit 3.			
Callee Number	Enter the caller name to filter the CDR report. CDR with the matching caller name will be filtered out.			
Source Trunk Name	Select source trunk(s) and the CDR of calls going through inbound trunk(s) will be filtered out.			
Destination Trunk Name	Select destination trunk(s) and the CDR of calls going outbound through the trunk(s) will be filtered out.			
Time	Specify the start time and the end time to filter the CDR report. Click on the calendar icon on the right and the calendar will shousers to select the exact date and time.			
Status	Filter with the call status, the available statuses are the following: o Answered o No Answer o Busy o Failed			

Table 52: CDR Filter parameters

The call report will display as the following figure shows.

Status \$	Call from \$	Call to \$	Start Time \$	Call Time \$	Talk Time ‡
0	"3100" 3100 [Trunk: Digital_1]	21007 [Trunk: sip147]	2018-11-13 18:52:14	14:09:17	14:09:15
•	"3100" 3100 [Trunk: Digital_1]	21006 [Trunk: sip147]	2018-11-13 18:51:52	0:00:09	0:00:05
•	"3100" 3100 [Trunk: Digital_1]	21007 [Trunk: sip147]	2018-11-13 18:51:03	0:00:36	0:00:22
•	"1007" 1007 [Trunk: sip147]	41000 [Trunk: Digital_1]	2018-11-13 15:29:57	0:00:26	0:00:07
•	"1007" 1007 [Trunk: sip147]	41000 [Trunk: Digital_1]	2018-11-13 15:28:31	0:00:08	0:00:00

Figure 91: Call Report

The CDR report has the following data fields:

Status

Answered, Busy, No answer, or Failed.

o Call From

Example format: "3100" 3100 [Trunk: Digital_1]

o Call To

Example format: 21007 [Trunk: sip147]

Start Time

Example Format: 2018-11-13 18:52:14

o Call Time

Example Format: 0:00:08

o Talk Time

Example Format: 0:00:07

CDR Report Operations

After applying the filter, Users could perform the following operations on the CDR report:

o Sort by data field

Click on the header of the data field column to sort the report according to an ascending or descending order. Clicking on the same header again reverse the order.

o Download the search result

Delete search result

o Delete all records

Click on The Delete All button to remove all the call report information.

o Download all records.

Click on L Download All Records to export all the records to a .csv file.

Automatic Download

Users could configure the GXW450X to automatically download the CDR records and send the records to an Email address. Click on "Automatic Download Settings" and configure the parameters in the dialog below.

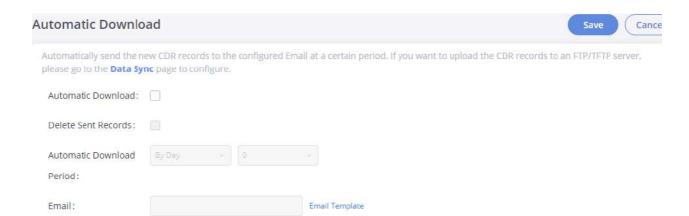


Figure 92: Automatic CDR Download

To receive CDR records automatically from Email, check "Enable" and select a time period "By Day" "By Week" or "By Month", and select Hour of the automatic download period. Make sure you have entered an Email or multiple email addresses where to receive the CDR records.

Users have the option to delete the sent records "Delete Sent Records".

CDR Report Data Fields

The CSV CDR report file downloaded will have the following data fields.

Field	Type	Description			
Account Code	String	An account code associated with the Party A channel.			
Caller Number	String	The Caller ID number.			
Callee Number	String	The destination number.			
Context	String	The context of the call.			
CallerID	String	The caller ID.			
Source Channel	String	The name of the source channel.			
Dest Channel	String	The name of the destination channel.			
Lastapp	String	The last application the Party A channel executed.			
Lastdata	String	The application data for the last application the Party A channel executed.			
Start time	Date/tim e	he time the CDR was created.			
Answer Time	Date/tim e	The time when Party A was answered, or when the bridge between Party A and Party B was created.			
End time	Date/tim e	The time when the CDR was finished. This occurs when either party hangs up, or when the bridge between the parties is broken.			
Call time	Integer	The time in seconds from start time until the end time.			
Talk time	Integer	The time in seconds from answer time until the end time.			

Field	Туре	Description	1	
Disposition	Enum	The final known disposition of the CDR record. The possible values are: "ANSWERED", "NO ANSWER, CONGESTION, FAILED, and BUSY.	ı	
Amaflags	Enum	A flag is specified on the Party A channel. The possible values are: "OMIT, BILLING, and DOCUMENTATION.	ı	
Uniqueid	String	A unique identifier for the Party A channel	ı	
Userfield	String	A user-defined field set on the channels. If set on both the Party A and Party B channels, the user fields of both are concatenated and separated by a comma.	ı	
Dest Channel Extension	String	The destination extension of the call	ı	
Caller Name	String	The name of the caller	ı	
Answer by	String	The extension to be called	ı	
Session	String	A numeric value that, combined with uniqueid and linkedid, can be used to uniquely identify a single CI record		
Action Owner	String	The party that made the call	Ī	
Action Type	String	The action type of the call	ı	
Source Trunk Name	Sting	The inbound route trunk name	ı	
Dest Trunk Name	String	The outbound route trunk name	ı	

Table 53: CDR Report Data Fields

Example of a CDR report Data fields:

Account code: —
 Caller Number: 1008
 Callee number: 1006
 Context: did-out
 Caller ID: "" <1008>

• Source Channel: DAHDI/i1-1-1

• **Dest Channel:** PJSIP/trunk_5-00000000

• Lastapp: Dial

 $\circ \ \textbf{Lastdata:} \ PJSIP/1006@trunk_5,, b(callee-handler^s^1)$

Start time: 11/13/2018 3:01:28 PM
 Answer time: 11/13/2018 3:01:31 PM
 End time: 11/13/2018 3:01:50 PM

o Call time: 22 (in seconds)

o Talk Time: 18

Disposition: ANSWEREDAmaflags: DOCUMENTATIONUniquelD: 1542092488

Userfield: External

○ **Dest channel extension:** trunk_5

○ Caller name: –

• **Answer by:** trunk_5

o Session: 1542092488529109-1008

Action owner: 1008Action type: DIAL.

Source Trunk name: Digital_1Dest Trunk name: sip147

CHANGE LOG

This section documents significant changes from previous versions of GXW450X user manuals. Only major new features or major document updat listed here. Minor updates for corrections or editing are not documented here.

Firmware version 1.0.1.11

- o Added new option "Original Called" when using SS7 signaling type. [Original Called]
- Added support for "early ACM" in SS7 interface settings. [Early ACM]
- Added new option PROGRESS to the Edit Digital Port. [PROGRESS]
- Added auth-user and auth-pass parameter support in OpenVPN. [User authentification]
- o Added ability to provision TR-069 settings. [TR-069]
- o Firewall rules will now apply to traffic between LAN and WAN interfaces. [Static Defense]
- Added E1/T1/J1 Error Code option to allow users to send either 480 or 503 response to the VoIP trunk when the E1/T1/J1 interface is down or unavailable. [E1/T1/J1 Error Code]

Firmware version 1.0.1.9

o No major changes

Firmware version 1.0.1.7

- o Added the ability to customize call failure tone settings. [Call Failure Tone Settings]
- Added ability to group E1 lines with a single DATA channel. [Data channel]
- Added the ability to assign channel groups. [Digital Hardware Configuration]
- o Added T1 E&M signaling support. [Table 30: Digital Hardware Configuration Parameters: T1 E&M Immediate]
- Added Secondary SIP Server option to Register SIP Trunks. [Secondary SIP Server]
- o Added Secondary Outbound Proxy option. [Backup Outbound Proxy]
- The administrator-level users can now view the SNMP page. [SNMP]
- Added GDMS support.

Firmware version 1.0.0.35

- o Added ability to select a different E1/T1 port to use for the line's data channel when SS7 is selected as the signaling type. [Data channel]
- o Added ability to select the minimum and maximum TLS versions to support. [TLS Security]
- Added SNMP support. [SNMP]
- Added options From User, Send PPI Header, PPI Mode, and DOD as From Name. [Table 33: VoIP Trunk Configuration Parameters Register SII [Table 34: VoIP Trunk Configuration Parameters Peer SIP Trunk]

Firmware version 1.0.0.29

- o Added the ability to configure tone region. [Tone Region]
- o Added the ability to configure DODs for VoIP trunks. [Direct Outward Dialing (DOD)]

Firmware version 1.0.0.27

- o Added GAPS provisioning support.
- o Added the Set Caller ID option to the Edit Inbound Rule page. [Set Caller ID Info]
- Added the Enable Filter on Source Caller ID option to the Edit Outbound Rule page. This option will allow only callers with the specified CID p to use the outbound route. [Enable Filter on Source Caller ID]

Firmware version 1.0.0.24

o No major changes.

Firmware version 1.0.0.22

o No major changes.

Firmware version 1.0.0.20

o No major changes.

Firmware version 1.0.0.18

o This is the initial version.

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