

Grandstream GXE502x IPPBX Quick Start Guide



Grandstream GXE502x IPPBX Quick Start Guide

This Quick Start Guide will quickly take you through the necessary steps to set up your Grandstream GXE502x. For advanced configuration instructions please consult the GXE user manual after completing the steps in this guide.

Check your GXE502x packaging:

- GXE502x unit
- Ethernet cable
- 12V DC power adapter

System Set-up Prerequisites:

- Compatible PC (to access web GUI)
- Ready access to power and UPS or surge protector
- DHCP must be configured with DHCP Option 66 enabled for Grandstream phones to be autoprovisioned (Grandstream default configuration. Non-Grandstream IP phones will need to be configured manually).

Connecting the GXE502x:

1. Connect one end of an RJ-45 Ethernet cable into the LAN port of the GXE.
2. Connect the other end of the Ethernet cable into the uplink port of an Ethernet switch/hub.

The GXE will be the router for this network, and will serve as a DHCP server as well. However, do not connect phones and other SIP devices to the switch/hub at this point; they will need to be connected later to be auto-provisioned.

3. Connect the 12V DC power adapter into the 12V DC power jack on the back of the GXE. Insert the main plug of the power adapter into a surge-protected power outlet.
4. Let the GXE boot up for the first time. You will know it is finished with the boot process when the "READY" LED is solid green.
5. Connect PSTN lines from the wall jack to the FXO ports, and analog lines (phone and fax) to the FXS ports. IP phones will be connected in a later step.

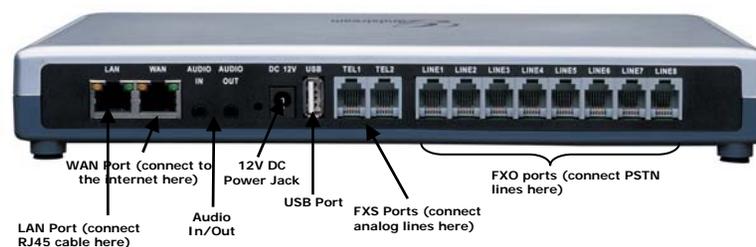
Note: Before configuring your GXE please go to <http://www.grandstream.com/BETATEST> for the newest firmware releases, voice prompts and updated GXE documentation.

Switch/Hub: network device used to share network connections

PSTN Line: traditional analog POTS line

DHCP: dynamic host configuration protocol – automatically assigns IP addresses

Extension Length: number of characters in an extension. For example: ext 803 has an extension length of 3.



GXE Configuration Steps

ONE. Access the web GUI (web configuration pages):

1. Connect a Windows PC to the same switch/hub that is connected to the GXE.
2. Click on the start menu and go to the "Run" option and type "cmd" to launch the Windows command prompt window.
3. Enter the following commands in the prompt window: *ipconfig /release* followed by *ipconfig/renew*. Press the enter button on your keyboard to run each command.
4. Open up a web browser of your choice (Microsoft Internet Explorer 7 recommended).
5. Type the default GXE LAN port IP address in the URL field: 192.168.10.1 and press the Enter key and the GXE web configuration login screen will load immediately. (See Figure 1-1)
6. The default login name and password are both "admin."
7. After logging in, the GXE web configuration interface (web GUI) will appear.



Figure 1-1

Note: All GUI screenshots are examples, do not copy these configurations into your GXE.

TWO: Set-up Wizard (Express Setup):

The set-up wizard is pre-configured with basic call settings that enable you to quickly configure your GXE. Use this wizard to begin making calls immediately and to understand the basic configuration set-up. **Note: The Express Setup Feature can only be run when the GXE is at the factory default settings.**

1. Click on the yellow **Express Setup** link at the left of the menu bar to load the Express Setup screen (See an example in Figure 2-1). This page lets you define the length of your extensions as well as the extension prefix and the identifier for your office (IP-PBX #).
2. Click "Next" to bring up the Automatic Extensions Provisioning page (See Figure 2-2 on the next page). This page shows a list of the default extension numbers for various key GXE extensions. Please ensure that the extension length and extension digit prefix matches the values set in the previous Express Setup page.

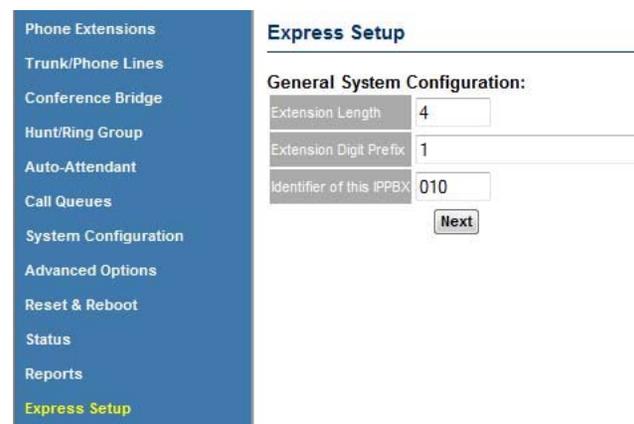


Figure 2-1

- Click on the button labeled: "Start Auto Provisioning" (See Figure 2-2). Once the Automatic Extensions Provisioning screen loads (See Figure 2-3), the GXE will begin finding devices by waiting for them to communicate via DHCP. At this point, please start connecting all SIP devices/phones into the switch/hub. The page will keep refreshing as the GXE looks for new devices.

In addition, your SIP devices/phones need to have **DHCP enabled** as well as have **DHCP Option 66** allowed to be auto-provisioned; these are factory default settings on Grandstream products as well as many others.

To ensure proper auto-provisioning of Grandstream IP phones, connect the IP phones to the switch/hub after Step 3, as the GXE will only auto-provision phones found during this stage.

- As the SIP devices/phones are auto-provisioned, the "Number of extensions provisioned" counter will increment to inform you of the auto-provisioning progress. Press the "Stop Auto Provisioning and Done" button when all devices are configured (See Figure 2-3). The Extensions Directory page will load automatically (See Figure 2-4). This will show you the default operator extension and all devices found during the Auto Provisioning step.
- You are now ready to make phone calls locally!! Try to make calls between your extensions.
- To make a call pick up the receiver and dial the extension and press the send button or #.

Express Setup

Automatic Extensions Provisioning:

Extension Number for Phone/FAX Ports	1990	for port 1;	1991	for port 2;
Extension Number for Paging	1992			
Extension Number for Auto-Attendant	1993			
Extension Number for Conference	1994	for conference 1;	1995	for conference 2;
Starting Extension For Express Provisioning	1000			
Ending Extension For Express Provisioning	1999			
Extension Number for Operator	1000			

Figure 2-2

Express Setup [Logout](#)

The auto provisioning process has now started and will continuously run until the "Stop Auto Provisioning And Done" button is pressed. Please connect the LAN port of the IP phones to the LAN port of the IPPBX, power them up and they will be provisioned within about 1 minute. Please only press the "Stop Auto Provisioning And Done" button after you finish with all the IP phones that are to be auto-provisioned.

Auto Provisioning of Extensions:

Starting Extension For Express Provisioning	800
Ending Extension For Express Provisioning	899
Number of extensions provisioned	1

Figure 2-3

Extensions Directory GREEN=local extension BLUE=remote extension [Logout](#)

All	Extension Name	Department	Device Type	IP Address	Status	Privilege	Action
<input type="checkbox"/>	800	Operator			Off-line	Super	<input type="button" value="Modify"/> <input type="button" value="Delete"/>
<input type="checkbox"/>	801		Grandstream GXP2000	1.1.4.18 192.168.2.100	Online	Regular	<input type="button" value="Modify"/> <input type="button" value="Delete"/>

Figure 2-4

THREE: Manually Adding Extensions

1. If you need to add more extensions manually, you can click on “Add One Extension” or “Batch Add” to add multiple extensions at a time. Choosing “Add One Extension” will load the “Add Extension page” (See Figure 3-1); enter a User Name, Extension, and SIP Password. You may set any of the other optional settings as you wish, then click the “Submit” button when done.
2. Next you will need to log into the web interface of the phone to be added by typing the IP address of the phone in your web browser.
3. Type in “admin” for the password on the phone’s login screen.
4. When you are logged in, the advanced settings page will load automatically.
5. Click on the Account 1 page of your phone to load the account configuration page (See Figure 3-2) and enter the 192.168.10.1 as the SIP Server.
6. Enter the corresponding information from the GXE’s “Add Extension” page for SIP User ID (The Extension #), Authenticate ID (The Extension#) and Authenticate password (SIP Password).
7. Scroll to the bottom of the account configuration page and click the “update” button. After the next page loads, click on the “reboot” button.
7. Congratulations, you have manually added an extension to your GXE!

The screenshot shows a web form titled "Add Extension". It contains several input fields and options: "User Name" (text input), "Department Name" (text input), "Extension" (text input with value "0000" and a dropdown for "800"), "Privilege" (dropdown menu with "Regular" selected), "SIP Password" (password input with asterisks), "Voicemail Allowed" (radio buttons for "Yes" and "No", "Yes" is selected), "Ring Attempts Forward to Voicemail" (dropdown menu with "4" selected), "Faxmail Allowed" (radio buttons for "Yes" and "No", "Yes" is selected), "Forward Voicemail/Faxmail to Email" (text input), "Password to Retrieve Voicemail/Faxmail" (text input with note "(0-9, #, cannot be void)"), "Call Forward" (radio buttons for "On" and "Off", "On" is selected), "Call Forward Phone" (text input), "Call Forward Rule" (dropdown menu with "None" selected), "Do Not Disturb" (radio buttons for "On" and "Off", "Off" is selected), and "Reboot Peer" (radio buttons for "Yes" and "No", "No" is selected). At the bottom are "Submit" and "Cancel" buttons.

Figure 3-1

The screenshot shows the "Grandstream Device Configuration" web page. At the top is a navigation bar with "STATUS", "BASIC SETTINGS", "ADVANCED SETTINGS", "ACCOUNT 1", "ACCOUNT 2", "ACCOUNT 3", "ACCOUNT 4", "ACCOUNT 5", and "ACCOUNT 6". The "ACCOUNT 1" tab is selected. Below the navigation bar, there are several configuration options: "Account Active" (radio buttons for "No" and "Yes", "Yes" is selected), "Account Name" (text input with example "(e.g., MyCompany)"), "SIP Server" (text input with example "(e.g., sip.mycompany.com, or IP address)"), "Outbound Proxy" (text input with example "(e.g., proxy.myprovider.com, or IP address)"), "SIP User ID" (text input with note "(the user part of an SIP address)"), "Authenticate ID" (text input with note "(can be same or different from SIP UserID)"), and "Authenticate Password" (text input with note "(not displayed for security protection)").

Figure 3-2

FOUR: Configuring the Network Settings

1. Click on "System Configuration" in the left hand column. This will automatically load the Network Settings configuration page (See Figure 4-1). On this page you can configure the LAN and WAN settings such as toggling between using DHCP or using a static IP address. You may choose to simply use the default setting values, or customize them if necessary.
2. The default LAN Base IP is 192.168.10.1/255.255.255.0. If you wish to use a different internal IP addressing scheme, please change it before running the express setup as the GXE will auto-provision the LAN Base IP into the phones for them to communicate with the GXE. Reboot the GXE before proceeding to the next step to make the LAN Base IP setting change take effect.

Warning: Do NOT change the LAN Base IP after auto-provisioning phones after the express setup, this will break the connection between your configured phones and the GXE.

The screenshot displays the 'Networking Setting' configuration page, divided into 'LAN Setting' and 'WAN Setting' sections.

LAN Setting:

- LAN Base IP: 192, 168, 10, 1 (base IP for the LAN port, default is 192.168.2.1)
- LAN Subnet Mask: 255, 255, 255, 0 (default is 255.255.255.0)
- DHCP Enable: Enable Disable
- Start of DHCP IP Pool: 100
- End of DHCP IP Pool: 199
- DHCP IP Lease Time: 120 (in units of hours, default is 120 hours or 5 days)

WAN Setting:

- Dynamically Assigned Via DHCP
- Use PPPoE
 - PPPoE Account ID: []
 - PPPoE Password: []
 - Preferred DNS server: 0, 0, 0, 0
- Statically Configured
 - IP Address: 192, 168, 0, 160
 - Subnet Mask: 255, 255, 255, 0
 - Default Router: 192, 168, 0, 1
 - Primary DNS: []
 - Secondary DNS: []

Device Mode: NAT Router Bridge

WAN Side Http/Telnet Access: No Yes

DMZ IP: [] [] [] []

DDNS Active: Enable Disable

DDNS ISP Type: 3domain.hk

Site Name: []

DDNS Account: []

DDNS Password: []

Port Forwarding:

WAN Port	LAN IP	LAN Port	Protocol
0	[]	0	UDP Only
0	[]	0	UDP Only
0	[]	0	UDP Only
0	[]	0	UDP Only
0	[]	0	UDP Only
0	[]	0	UDP Only
0	[]	0	UDP Only
0	[]	0	UDP Only

Figure 4-1

FIVE: Configuring the System Settings

1. Click on "System Configuration" in the menu bar on the left.
2. Click on "System Settings" in the expanded menu to load the system settings page (See Figure 5-1). This page lets you configure the login and contact information of the system administrator, email options, time zones, call record servers and storage quotas.
3. For better security, please change the web configuration password by entering your own password in the "Login Password" field. Keep this password safe and do not lose it.
4. To enable the voicemail-to-email feature, please create a new email address or use an existing one.
5. Enter the email account SMTP Server, Login Name, Login Password, and Email Address settings in the corresponding fields; you may obtain this information from your email provider.
6. Click on the "Submit" button to save your changes.

System Setting

Administrator Setting

Login Password	<input type="text"/>
Name	<input type="text"/>
Contact Phone	<input type="text"/>
Contact Mobile	<input type="text"/>
Contact Email	<input type="text"/>

System Name	<input type="text"/>
SIP Port	<input type="text"/>
Static Mapped WAN IP for SIP	<input type="text"/>
Static Mapped WAN Port for SIP	<input type="text"/>
STUN Server	<input type="text"/> (e.g., my_stunserver_ip_or_urt:port)
SMTP Server	<input type="text"/>
Login Name	<input type="text"/>
Login Password	<input type="text"/>

Email Address	<input type="text"/>
CDR TFTP Server	<input type="text"/>
Time Zone	GMT-5:00 (US Eastern Time, New York)
Self-Defined Time Zone	<input type="text"/> (For example: "MTZ+6MDT+5.M4.1.0.M11.1.0")
NTP Server	<input type="text"/>
Firmware Upgrade Server	fw.grandstream.com/gs No implement yet
Provisioning Server	fw.grandstream.com/gs No implement yet
Syslog Server IP	<input type="text"/>
Syslog Server Port	<input type="text"/>

Storage quota of voicemail/voicemail/faxmail per privilege level

super	2%	▼
Privileged	2%	▼
Regular	2%	▼
Basic	2%	▼
Restricted	2%	▼

Figure 5-1

SIX: Configuring PSTN Lines (if available)

1. Click on TRUNK/Phone Lines on the left menu bar to load the Internal PSTN Trunk Line configuration page (See Figure 6-1).
2. All sections on this page except for the Line Call Control section are used to match the settings of the GXE to the local PSTN line parameters. The default settings should work in most cases; you may check with your PSTN service provider if your line parameters differ so that these settings may be adjusted accordingly.
3. Scroll down to the bottom of the page to view the "Line Call Control" Settings (Figure 6-2). To direct all incoming PSTN calls on all lines to an operator, for example, first enter "1-8" in the Line field.
4. In the Inbound Call Answer drop-down box, choose "Direct Extension." In the drop-down box to the right, choose the extension number of the operator.
5. Set the Dial Prefix field to the prefix used to for outgoing PSTN calls; for example, "9" is commonly used.

Note: This section is very important as it configures how all incoming calls are handled on the FXO lines. For example you could have all incoming calls go to the auto-attendant for one line and add a dial pre-fix for another.

FXO Termination

Enable Current Disconnect	<input type="radio"/> No <input checked="" type="radio"/> Yes
Current Disconnect Threshold(ms)	200 (100-10000, default 200)
Enable Tone Disconnect	<input checked="" type="radio"/> No <input type="radio"/> Yes
Polarity Reversal Disconnect	<input checked="" type="radio"/> No <input type="radio"/> Yes
AC Termination Impedance	USA

Call Progress Tones (Syntax: f1=val@vol,f2=val@vol,c=on1/off1/on2/off2/on3/off3;[...])

Dial Tone	f1=350@-13,f2=440@-13,c=1000
Ring Back Tone	f1=440@-19,f2=480@-19,c=2000/4000
Busy Tone	f1=480@-24,f2=620@-24,c=500/500

Line Dialing Settings

DTMF Digit Length(ms)	ch1-8:80	(Example: ch1-8:35;)
DTMF Digit Volume(dBm)	ch1-8:-11	(Example: ch1-8:-11;)
DTMF Dial Pause(ms)	ch1-8:100	(Example: ch1-8:35;)
Wait for Dial-Tone(Y/N)	ch1-8:N	(Example: ch1-8:Y;)
Minimum Delay Before Dial	ch1-8:1000	(Example: ch1-8:100;)

Caller ID Standard

Caller ID Scheme	Bellcore
------------------	----------

Figure 6-1

Line Call Control

1. Line	(e.g., 1-4,7)	Dial Prefix		Inbound Call Answer	Auto-Attendant	None	Delete
2. Line	(e.g., 1-4,7)	Dial Prefix		Inbound Call Answer	Auto-Attendant	None	Delete
3. Line	(e.g., 1-4,7)	Dial Prefix		Inbound Call Answer	Auto-Attendant	None	Delete

Allow Other PPBX Use: None

Submit Cancel

Figure 6-2

SEVEN: Configuring SIP Trunks (if available)

1. Click on TRUNK/Phone Lines on the left menu bar and then click on "SIP Trunk" to load the SIP Trunk configuration page (**Figure 7-1**).
2. Your SIP service provider will provide you with SIP account and registration information, which you may enter in the respective fields of this page.
3. Set the "Trunk Active" setting to "Enable" to enable this trunk on your GXE. You may also specify an arbitrary name for this trunk in the "Trunk Name" field, as well as one for the SIP account in the "Account Name" field.
4. To set a limit on the number of calls allowed through this trunk, please set the "Max Concurrent Calls Allowed" field to the limit allowed by your service provider, or a lower number if you wish.
5. Set the dial digit prefix which users will prefix outbound numbers with to specify this trunk to be used.
6. Lastly, use the "Inbound Call Answer" setting to specify the inbound call destination, such as "auto-attendant".

Trunk Name	
Trunk Active	<input type="radio"/> Enable <input checked="" type="radio"/> Disable
SIP Server URL	
Outbound Proxy URL	
Account Name	
Account ID	
Authenticate ID	
Password	
Registration Retry Interval	600 second(s)
Heart Beat	<input checked="" type="radio"/> Yes <input type="radio"/> No
Max Concurrent Calls Allowed	8
Dial Prefix	
Inbound Call Answer	Auto-Attendant <input type="button" value="None"/>
<input type="button" value="Submit"/> <input type="button" value="Cancel"/>	

Figure 7-1

EIGHT: Configuring a Peer System (if available)

1. If you have more than one GXE502x, you can configure the units to be peers. To do this, click on "Advanced Options" in the menu bar to load the "Peer Systems" page.
2. Click the "Add" button to load the "Add Peer System" Page (**Figure 8-1**). This page lets you enter configuration data for the remote GXE you are adding as a peer system.
3. In the Peer URL field, enter the IP address or domain name of the peer GXE. Assign a number to the Trunk Number parameter to set the # of concurrent calls possible on the peer system.
4. Set the extension prefix of the remote PBX extensions in the Prefix field. The GXE will route calls starting with this prefix to the peer system.
5. Assign a number to this peer system in the IPPBX Number field.
6. If the peer system is another GXE, set the Heart Beat setting to "Yes." Otherwise, use "No."
7. Reboot the GXE so the configuration changes can take effect.

Peer URL	67.153.142.125
Trunk Number	010
Prefix	6 (up to 10 digits)
IPPBX Number	006
Heart Beat	<input checked="" type="radio"/> Yes <input type="radio"/> No
<input type="button" value="Submit"/> <input type="button" value="Cancel"/>	

Figure 8-1