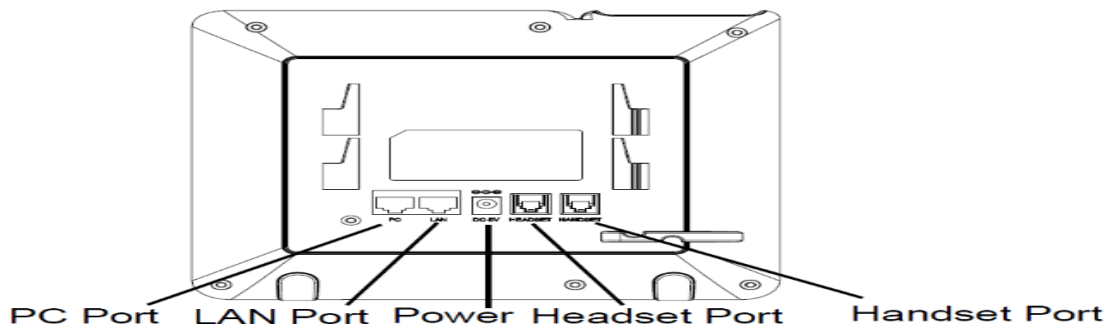
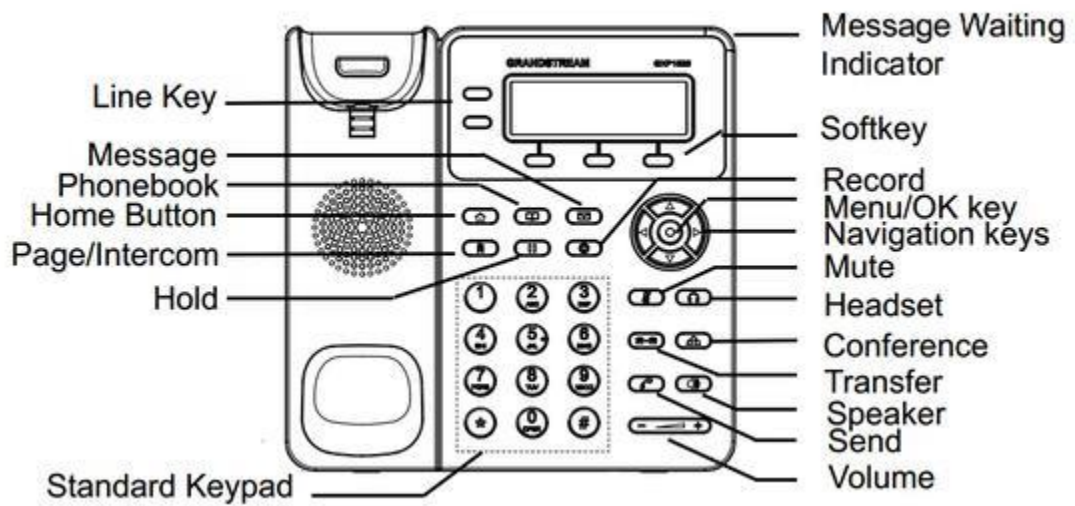


GRANDSTREAM IP PHONE GXP16XX

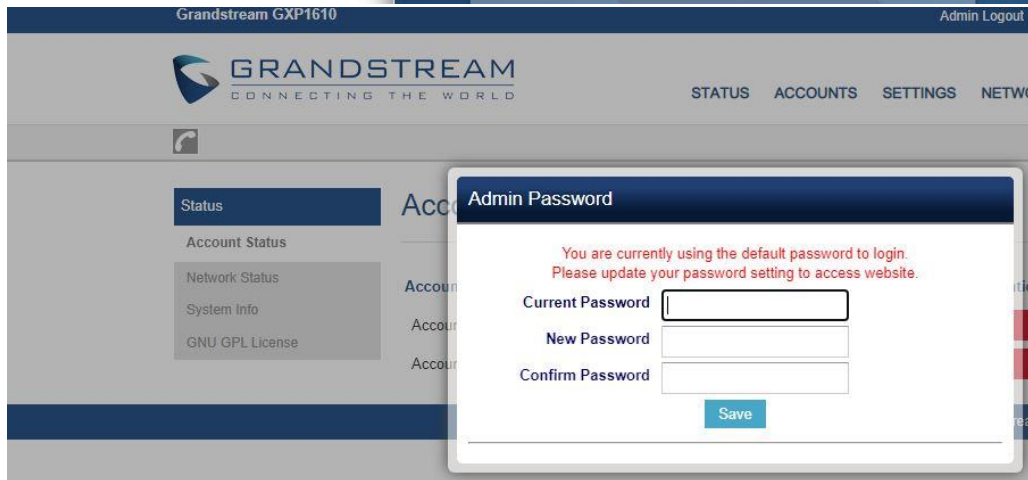
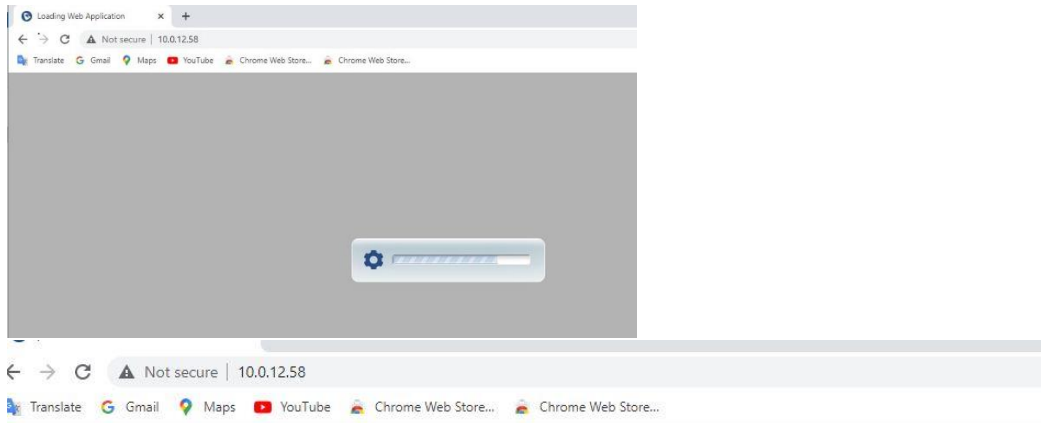


1. Интернет сүлжээн холбож тогонд залган асаана
2. Интернет утас нь ассаны дараа автоматаар тухайн сүлжээний интернет хаяг авах бөгөөд аппаратны нүүрэн хэсэгт байрлах **Menu/Ok Key Navigation keys** ийн **Ok** товчин дээр **драхад Status- network орж IPV4:xx.xxx.xxx.xx хаяг авсан байна**



3. Тухайн сүлжээнээс Web ээр дээрхи авсан хаягаар орно

GRANDSTREAM IP PHONE GXP16XX



4.GXS1610 grandsteam Web ээр хандан орох default

User name:admin

Password:admin

Хийн нэвтрэхэд password солих цонх гарж ирэн password солин нэвтрэн орно

GRANDSTREAM IP PHONE GXP16XX

GXP1610 Admin Logout | Reboot | Factory Reset English

GRANDSTREAM
CONNECTING THE WORLD

STATUS ACCOUNTS SETTINGS NETWORK MAINTENANCE CONTACTS

Version

Account Status

Account	SIP User ID	SIP Server	SIP Registration
Account 1			NO
Account 2			NO

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Grandstream GXP1610 Admin Logout | Reboot | Factory Reset English

GRANDSTREAM
CONNECTING THE WORLD

STATUS **ACCOUNTS** SETTINGS NETWORK MAINTENANCE CONT.

Account 1 General Settings
Account 2 Network Settings

Version 1.1

Account Status

Account	SIP User ID	SIP Server	SIP Registration
Account 1			NO
Account 2			NO

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GRANDSTREAM IP PHONE GXP16XX

General Settings

Network Settings

SIP Settings

Audio Settings

Call Settings

Intercom Settings

Feature Codes

Account 2

Account Active No Yes

Account Name 70086008

SIP Server 202.179.20.22

Secondary SIP Server

Outbound Proxy 202.179.20.22

Backup Outbound Proxy

BLF Server

SIP User ID 70086008

Authenticate ID 70086008

Authenticate Password

Name 70086008

Voice Mail Access Number

Account Display User Name User ID

Save Save and Apply Reset

Accounts

Account 1

General Settings

Network Settings

SIP Settings

Audio Settings

Call Settings

Intercom Settings

Feature Codes

Account 2

Audio Settings

Send DTMF in-audio via RTP (RFC2833) via SIP INFO

DTMF Payload Type 101

Preferred Vocoder - choice 1 PCMA

Preferred Vocoder - choice 2 PCMU

Preferred Vocoder - choice 3 PCMA

Preferred Vocoder - choice 4 PCMU

Preferred Vocoder - choice 5 PCMA

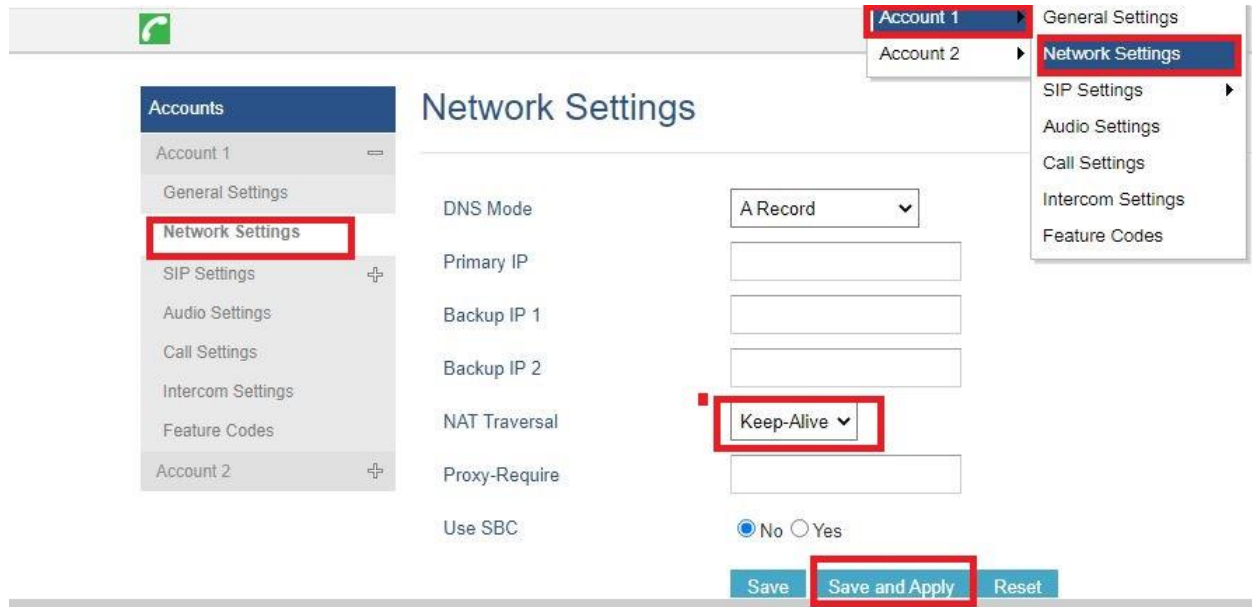
Preferred Vocoder - choice 6 PCMU

Preferred Vocoder - choice 7 PCMA

Use First Matching Vocoder in 200 OK No Yes

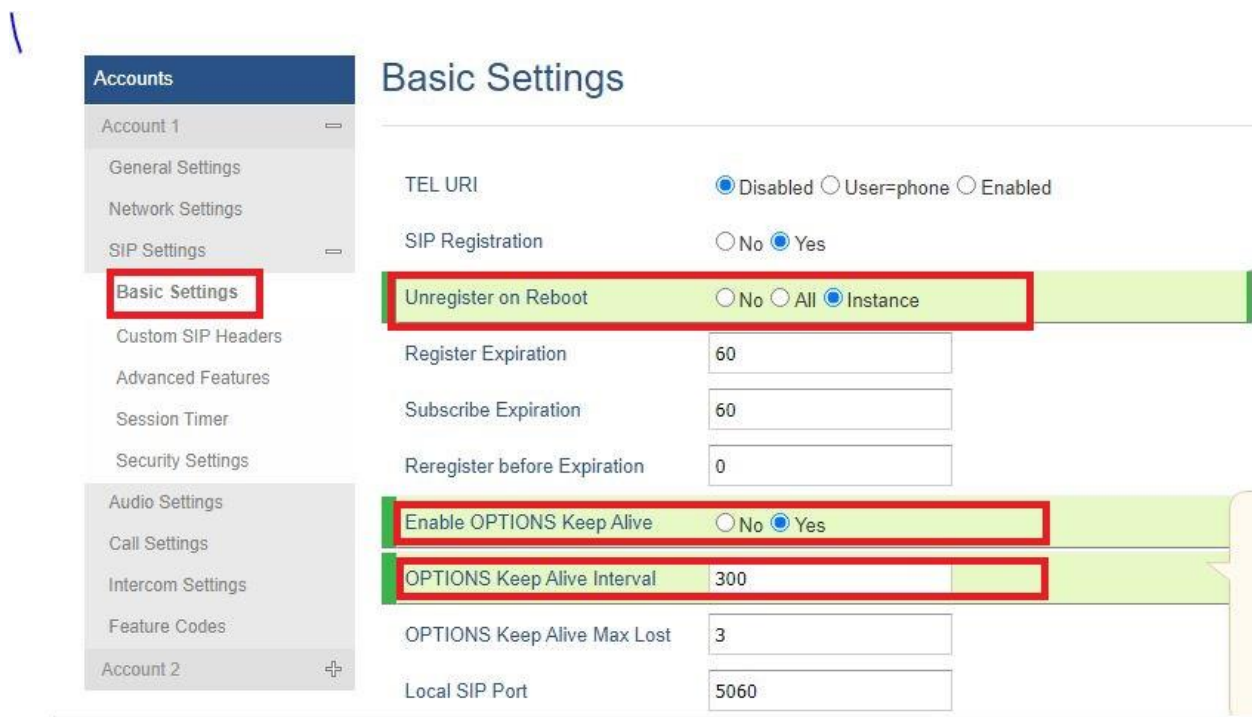
GRANDSTREAM IP PHONE

GXP16XX



The screenshot shows the 'Network Settings' page for Account 1. The left sidebar contains a menu with 'Network Settings' highlighted. The top right shows a dropdown menu with 'Network Settings' selected. The main content area includes fields for DNS Mode (A Record), Primary IP, Backup IP 1, Backup IP 2, NAT Traversal (Keep-Alive), Proxy-Require, and Use SBC (No/Yes). At the bottom are 'Save', 'Save and Apply', and 'Reset' buttons.

Field	Value
DNS Mode	A Record
Primary IP	
Backup IP 1	
Backup IP 2	
NAT Traversal	Keep-Alive
Proxy-Require	
Use SBC	<input checked="" type="radio"/> No <input type="radio"/> Yes



The screenshot shows the 'Basic Settings' page for Account 1. The left sidebar contains a menu with 'Basic Settings' highlighted. The main content area includes fields for TEL URI (Disabled/Enabled), SIP Registration (No/Yes), Unregister on Reboot (No/All/Instance), Register Expiration (60), Subscribe Expiration (60), Reregister before Expiration (0), Enable OPTIONS Keep Alive (No/Yes), OPTIONS Keep Alive Interval (300), OPTIONS Keep Alive Max Lost (3), and Local SIP Port (5060). Several fields are highlighted with red boxes.

Field	Value
TEL URI	<input checked="" type="radio"/> Disabled <input type="radio"/> User=phone <input type="radio"/> Enabled
SIP Registration	<input type="radio"/> No <input checked="" type="radio"/> Yes
Unregister on Reboot	<input type="radio"/> No <input type="radio"/> All <input checked="" type="radio"/> Instance
Register Expiration	60
Subscribe Expiration	60
Reregister before Expiration	0
Enable OPTIONS Keep Alive	<input type="radio"/> No <input checked="" type="radio"/> Yes
OPTIONS Keep Alive Interval	300
OPTIONS Keep Alive Max Lost	3
Local SIP Port	5060

GRANDSTREAM IP PHONE GXP16XX

The screenshot shows the Grandstream web interface. At the top, there is a navigation bar with the logo and the text "GRANDSTREAM CONNECTING THE WORLD". To the right of the logo are menu items: STATUS, ACCOUNTS, SETTINGS, NETWORK, MAINTENANCE, and CONTACT. Below the navigation bar, there is a dropdown menu for "Account 1" and "Account 2". The "Account 1" dropdown is open, showing a list of settings: General Settings, Network Settings, SIP Settings, Audio Settings, Call Settings, Intercom Settings, and Feature Codes. The "SIP Settings" dropdown is also open, showing: Basic Settings, Custom SIP Headers, Advanced Features, Session Timer, and Security Settings. The "Security Settings" option is highlighted with a red box. In the main content area, the "Security Settings" page is displayed for "Account 1". The left sidebar shows a list of settings: General Settings, Network Settings, SIP Settings, Basic Settings, Custom SIP Headers, Advanced Features, Session Timer, and Security Settings. The "Security Settings" page contains several options with radio buttons: "Check Domain Certificates" (No selected), "Validate Certificate Chain" (No selected), "Validate Incoming Messages" (No selected), "Check SIP User ID for Incoming INVITE" (No selected), and "Accept Incoming SIP from Proxy Only" (Yes selected). The "Accept Incoming SIP from Proxy Only" option is highlighted with a red box.

The screenshot shows the Grandstream web interface for "Account 1". The left sidebar shows a list of settings: General Settings, Network Settings, SIP Settings, Audio Settings, Call Settings, Intercom Settings, and Feature Codes. The "Call Settings" option is highlighted with a red box. The main content area displays the "Call Settings" page. The settings include: "Early Dial" (No selected), "Dial Plan Prefix" (empty text box), "Dial Plan" (text box containing "{ x+ | *x+ | **x*x+ }"), "Bypass Dial Plan" (checkboxes for Contact, Call History Incoming Call, Call History Outgoing Call, and Dialing Page, all unchecked), "MPK" (checkbox checked), "API" (checkbox unchecked), "Call Log" (radio buttons for Log All Calls selected, Log Incoming/Outgoing Only (missed calls NOT recorded), and Disable Call Log), "Send Anonymous" (No selected), "Anonymous Call Rejection" (Yes selected), "Auto Answer" (No selected), "Auto Answer Numbers" (empty text box), and "Refer-To Use Target Contact" (No selected). The "Anonymous Call Rejection" option is highlighted with a red box.

GRANDSTREAM IP PHONE GXP16XX

The screenshot shows the Grandstream web interface. At the top, there is a navigation bar with 'STATUS', 'ACCOUNTS', 'SETTINGS', 'NETWORK', and 'M'. The 'SETTINGS' menu is open, showing options like 'General Settings', 'Call Features', 'Multicast Paging', 'Preferences', 'Web Service', 'XML Applications', 'Programmable Keys', 'Broadsoft XSI', and 'External Service'. The 'Call Features' option is highlighted. Below the navigation, the 'Call Features' configuration page is visible. It includes a sidebar with 'Settings', 'General Settings', 'Call Features', 'Multicast Paging', 'Preferences', and 'Web Service'. The main content area shows several settings: 'Onhook Dial Barging' (radio buttons for Disabled and Enabled, with Enabled selected), 'Off-hook Auto Dial' (text input field), and 'Off-hook Auto Dial Delay' (text input field with '0').

This screenshot shows a list of call feature settings. Each setting has a label and a control element (text input or radio buttons). The 'Disable Direct IP Call' setting is highlighted with a red box. The settings are: 'Intercom Multicast Paging Address' (text input), 'Intercom Multicast Paging Label' (text input), 'Bypass Dial Plan Through Call History and Directories' (radio buttons for No and Yes, with No selected), 'Disable Call Waiting' (radio buttons for No and Yes, with No selected), 'Disable Call Waiting Tone' (radio buttons for No and Yes, with No selected), 'Ring For Call Waiting' (radio buttons for No and Yes, with No selected), 'Disable Busy Tone on Remote Disconnect' (radio buttons for No and Yes, with No selected), 'Disable Direct IP Call' (radio buttons for No and Yes, with Yes selected), 'Use Quick IP Call Mode' (radio buttons for No and Yes, with No selected), and 'Disable Conference' (radio buttons for No and Yes, with No selected).

The screenshot shows the 'Account Status' page in the Grandstream web interface. The browser address bar shows '10.0.12.58/#page:status_account'. The navigation bar includes 'Grandstream GXP1610', 'Admin Logout', 'Reboot', 'Factory Reset', and 'English'. The 'Account Status' page has a sidebar with 'Status', 'Network Status', 'System Info', and 'GNU GPL License'. The main content area shows a table with columns for 'Account', 'SIP User ID', 'SIP Server', and 'SIP Registration'. The 'SIP Registration' column has a green 'YES' button for 'Account 1' and a red 'NO' button for 'Account 2'. The 'Account Status' menu item in the navigation bar is highlighted with a red box.

Account	SIP User ID	SIP Server	SIP Registration
Account 1	70086008	202.179.20.22	YES
Account 2			NO