

## **GXP1610/1615/1620/1625/1628/1630 Firmware Release Note**

### **IMPORTANT UPGRADING NOTE**

- Starting from firmware version 1.0.2.21, downgrading to version 1.0.2.4 or lower requires a matched protocol for upgrading and provisioning under web UI→Maintenance→Upgrade and Provision→option “Config Upgrade Via” and option “Firmware Upgrade Via”.

For example, if you are using HTTP server to downgrade to 1.0.2.4, please configure both “Config Upgrade Via” and “Firmware Upgrade Via” to “HTTP”.

- Firmware version 1.0.4.50 supports new hardware revision. For GXP16xx that's on new hardware version 2.2A cannot downgrade to firmware version lower than 1.0.4.50.

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## **FIRMWARE VERSION 1.0.4.50**

### **PRODUCT NAME**

GXP1610, GXP1615, GXP1620, GXP1625, GXP1628 and GXP1630

### **DATE**

01/20/2017

### **BUG FIX**

- Fixed Participants could not hear host after host presses RecOnsoftkey in conference call using G.723/iLBC
- Fixed Phone crashes with core dump generated when the call is Picked up with BLF key
- Fixed NTP default value issue

## FIRMWARE VERSION 1.0.4.46

### PRODUCT NAME

GXP1610, GXP1615, GXP1620, GXP1625, GXP1628 and GXP1630

### DATE

12/30/2016

### ENHANCEMENT

- Added support for concurrent G.729 calls
- Added option to enable/disable webUI access [ENABLE/DISABLE WEBUI ACCESS]
- Added option to hide the password on the display after dialing a specific prefix [HIDE PASSWORD DISPLAY AFTER DIALING CERTAIN PREFIX]
- Added LDAPS support [LDAPS SUPPORT]
- Added support for certificate DN validation for LDAPS
- Added option to remove SIP error on LCD [ABILITY TO REMOVE SIP ERROR ON LCD DISPLAY]

### BUG FIX

- Fixed Phone GUI crashed after checking Broadsoft directory
- Fixed Phone should wait to send REFER until the hold process is complete when doing call park
- Fixed Phone doesn't update the "To" header tag when sending PRACK message for provisional reliability
- Fixed Phone sends multiple DTMF tones when in-audio method
- Fixed Phone does not provide dial tone when doing a Conference
- Fixed Transfer hard-key should work even if auto-attended transfer isn't enabled
- Fixed Phone rejects call with 488 when there is incoming call from IVR using certain soft switch
- Fixed Phone ring timeout when transferring-no-answer is too short which is 30 seconds

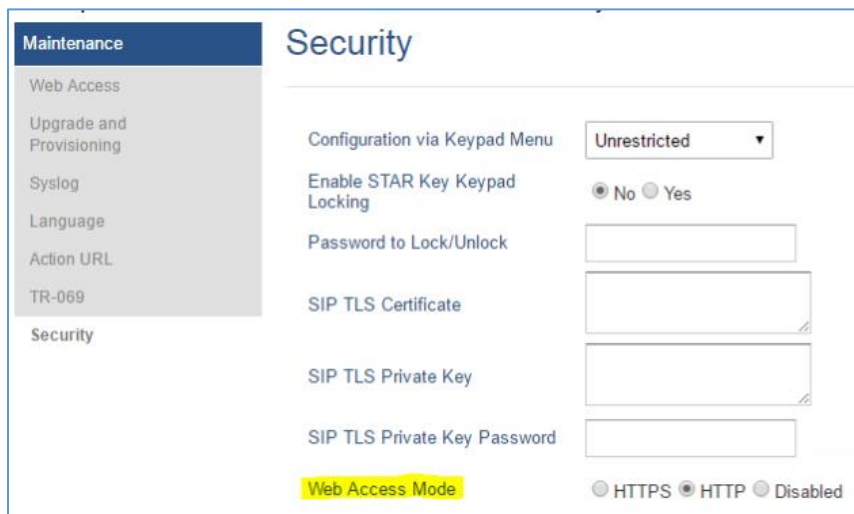
### NEW FEATURES OVERVIEW

This section lists major new features and describes how to use it from the user's point of view.

#### ENABLE/DISABLE WEBUI ACCESS

- **Web Configuration**

This option can be found under web UI → Maintenance → Security.



- **Functionality**

This option allows user to enable or disable web access for the phone.

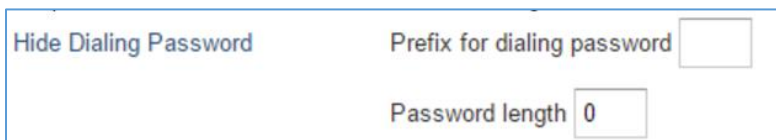
- **New P Value**

Pvalue	Description	Value Range	Default
P1650	Web Access Mode	Value = 0; HTTPS Value = 1; HTTP Value = 2; Disabled	1

## HIDE PASSWORD DISPLAY AFTER DIALING CERTAIN PREFIX

- **Web Configuration**

This option can be found under web UI → Account x → Call settings.



- **Functionality**

This option allows user to hide password from displayed on LCD after dialing certain prefix.

- **New P Value**

Pvalue	Description	Value Range	Default
P26049	Account 1: Prefix for dialing password		



P26050	Account 1: Password length		0
P26149	Account 2: Prefix for dialing password		
P26150	Account 2: Password length		0
P26249	Account 3: Prefix for dialing password		
P26250	Account 3: Password length		0

## LDAPS SUPPORT

- Web Configuration**

This option can be found under web UI → Phonebook → LDAP. Users can select “LDAP Protocol” as “LDAPS”.

### LDAP

LDAP protocol

LDAPS ▼

Server Address

Port

389

Base

User Name

Password

LDAP Number Filter

LDAP Name Filter

LDAP Version

☐ Version 2
 ☒ Version 3

LDAP Name Attributes

LDAP Number Attributes

LDAP Display Name

- Functionality**

This option allows user to select LDAPS as the LDAP protocol.

- New P Value**

Pvalue	Description	Value Range	Default
P8370	LDAP Protocol	Value = 0; LDAP Value = 1; LDAPS	0

## ABILITY TO REMOVE SIP ERROR ON LCD DISPLAY

- **Web Configuration**

This option can be found under web UI → Settings → Call Features.

Show SIP Error Response

☐ No
 ☒ Yes

- **Functionality**

This option allows user to enable or disable SIP error response to be displayed on LCD.

- **New P Value**

Pvalue	Description	Value Range	Default
P8376	Show SIP error response	Value = 0; No Value = 1; Yes	1

## FIRMWARE VERSION 1.0.4.33

### PRODUCT NAME

GXP1610, GXP1615, GXP1620, GXP1625, GXP1628 and GXP1630

### DATE

11/21/2016

### ENHANCEMENT

- Added support of Intercom Button to Send Multicast Paging [INTERCOM BUTTON]
- Added option to disable custom SIP Header [OPTION TO DISABLE CUSTOM SIP HEADER]
- Added support to send SIP log without enabling debug level [SUPPORT TO SEND SIP LOG WITHOUT ENABLING DEBUG LEVEL]
- Added option to bypass firmware upgrade confirmation [OPTION TO BYPASS FIRMWARE UPGRADE INFORMATION]
- Added support of confirmation message to reboot the phone [CONFIRMATION MESSAGE TO REBOOT THE PHONE]
- Added support to remember the headset mode after being set and reboot [SUPPORT TO REMEMBER THE HEADSET MODE AFTER BEING SET AND REBOOT]

### BUG FIX

- Fixed the wrong default value of Predictive Dialing Feature issue
- Fixed device GUI crash after pressing Intercom Key with empty Intercom Multicast Paging Address
- Fixed the Chinese confirmation message of reboot is not friendly
- Fixed Use X-Grandstream-PBX Header is invalid on some models
- Fixed SIP Log sent through syslog are truncated
- Fixed web UI works abnormally after setting Maximum Transmission Unit

### NEW FEATURES OVERVIEW

This section lists major new features and describes how to use it from the user's point of view.

#### INTERCOM BUTTON

- **Web Configuration**

This option can be found under web UI →Settings→ Call Settings.

**Intercom Key Mode**
☒ Intercom
☐ Multicast Paging

**Intercom User ID**

**Intercom Multicast Paging Address**

**Intercom Multicast Paging Label**

- **Functionality**

Allow users to port over the Multicast Paging mode that exists under Programmable Keys to the Intercom hard button.

- **New P Value**

Pvalue	Description	Value Range	Default
P22123	<New Pvalue> Intercom Key Mode	Value = 0; Intercom Value = 1; Multicast Paging	Intercom

### OPTION TO DISABLE CUSTOM SIP HEADER

- **Web Configuration**

This option can be found under web UI →Account X→SIP Settings→Customer SIP Headers.

**Use Privacy Header**
☒ Default
☐ No
☐ Yes

**Use P-Preferred-Identity Header**
☒ Default
☐ No
☐ Yes

**Use P-Access-Network-Info Header**
☐ No
☒ Yes

**Use P-Emergency-Info Header**
☐ No
☒ Yes

- **Functionality**

Allow users to manually enable/disable the SIP Header messages.

- **New P Value**

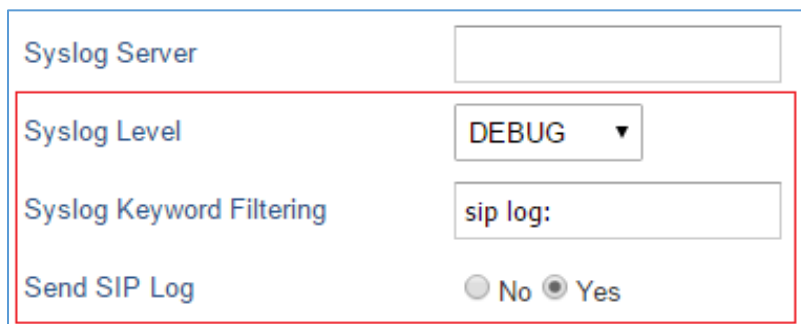
Pvalue	Description	Value Range	Default
P26058	<New Pvalue>	Value = 0; No	Yes

	Use P-Access-Network-Info Header	Value = 1; Yes	
P26059	<New Pvalue> Use P-Emergency-Info Header	Value = 0; No Value = 1; Yes	Yes

## SUPPORT TO SEND SIP LOG WITHOUT ENABLING DEBUG LEVEL

- Web Configuration**

This option can be found under web UI →Maintenance→Syslog.



- Functionality**

By setting Send SIP Log to Yes and the syslog to DEBUG then set the filter keywords to: “sip log:”, you will be able to receive only sip log and not all syslog messages.

- New P Value**

P value	Description	Value Range	Default
P208	Syslog Level	Value = 0; NONE Value = 1; DEBUG Value = 2; INFO Value = 3; WARNING Value = 4; ERROR	NONE
P22129	Syslog Keyword Filtering	STRING	N/A
P1387	SIP Log	Value = 0; No Value = 1; Yes	No

## OPTION TO BYPASS FIRMWARE UPGRADE INFORMATION

- Web Configuration**

This option can be found under web UI →Maintenance→Upgrade and Provisioning.

Firmware Upgrade Confirmation    ☐ No    ☒ Yes

- **Functionality**

Allow users to automatically upgrade to the version that the server has provided without needing to confirm the upgrade to avoid end users manually select no when doing mass upgrade.

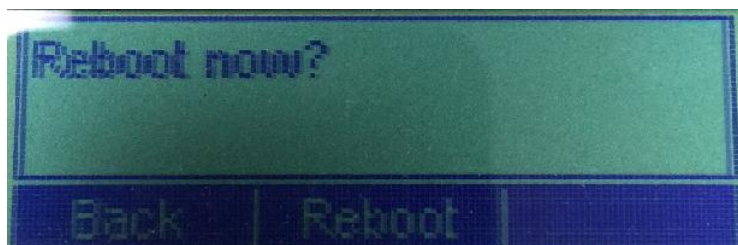
- **New P Value**

Pvalue	Description	Value Range	Default
P8375	Firmware Upgrade Confirmation	Value = 0; No Value = 1; Yes	Yes

#### CONFIRMATION MESSAGE TO REBOOT THE PHONE

- **Functionality**

Allow users to confirm reboot to prevent wrong clicking which might cause unnecessary.



#### SUPPORT TO REMEMBER THE HEADSET MODE AFTER BEING SET AND REBOOT

- **Functionality**

The phone now keeps the headset mode on regardless of power off or reboot.

## **FIRMWARE VERSION 1.0.4.22**

### **PRODUCT NAME**

GXP1610, GXP1615, GXP1620, GXP1625, GXP1628 and GXP1630

### **DATE**

10/03/2016

### **ENHANCEMENT**

- Added ability to perform a quick search and create new contact using Czech characters
- Added ability to perform a quick search and create new contact using Russian characters
- Added ability to lock settings on phone
- Added support for DHCP option 132 and 133 tunneled through DHCP option 43
- Added DHCP option 132 for (802.1Q VLAN ID) and 133 (QoS priority level)
- Added ability to adjust ringtone level from web GUI
- Added support for separated QoS setting for SIP and RTP
- Added configurable web access port
- Added ability to configure record button
- Optimized Broadsoft N-Way Conference

### **BUG FIX**

- Fixed blind transfer ignores the dial plan
- Fixed call forward using forward softkey ignores the dial plan.
- Fixed local phonebook starts duplicating contacts and sometimes not showing all contacts while Phonebook reaches 10 contacts
- Fixed DUT sends a BYE SIP message right after the INVITE when configured "DND Call Feature On"
- Fixed display of name override number on phonebook display
- Fixed phone does not send authorization header when receive 302 message
- Fixed phone cannot download provisioning file through 301 HTTP redirection
- Fixed phone is not reacting to Speed Dial SoftKey
- Fixed missing symbol slash "/" on phone keypad
- Fixed network connectivity lost when changing the Layer 2 QoS 802.1p Priority.
- Fixed GUI crashed after setting Dial Plan Prefix to a long Chinese string
- Fixed "Firmware Upgrade via" doesn't take effect after it is set via SSH2
- Fixed phone answer paging call even if custom call-info is configured

## NEW FEATURES OVERVIEW

This section lists major new features and describes how to use it from the user's point of view.

### QUICK SEARCH AND CREATE NEW CONTACT USING CZECH CHARACTERS

- **Web Configuration**

This option can be found under device web UI → Maintenance → Language → Select Czech language.

- **Functionality**

Search contact names with Czech characters.

### QUICK SEARCH AND CREATE NEW CONTACT USING RUSSIAN CHARACTERS

- **Web Configuration**

This option can be found under device web UI → Maintenance → Language → Select Russian language.

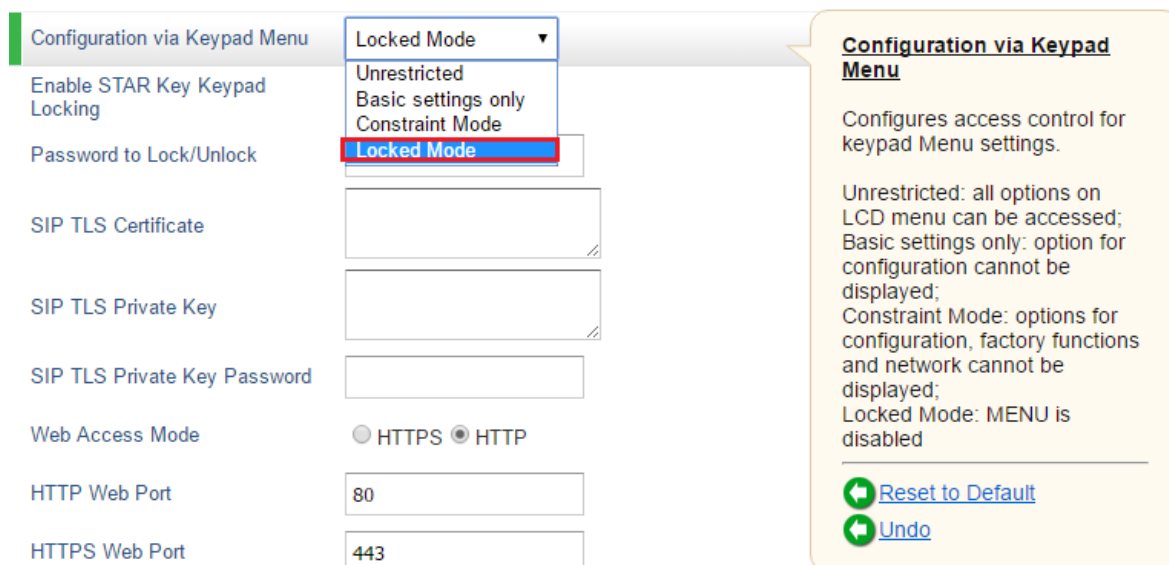
- **Functionality**

Search contact names with Russian characters.

### LOCK SETTINGS ON PHONE

- **Web Configuration**

This option can be found under device web UI → Maintenance → Security.



Configuration via Keypad Menu	Locked Mode ▼
Enable STAR Key Keypad Locking	Unrestricted
Password to Lock/Unlock	Basic settings only
	Constraint Mode
	<b>Locked Mode</b>
SIP TLS Certificate	
SIP TLS Private Key	
SIP TLS Private Key Password	
Web Access Mode	<input type="radio"/> HTTPS <input checked="" type="radio"/> HTTP
HTTP Web Port	80
HTTPS Web Port	443

**Configuration via Keypad Menu**

Configures access control for keypad Menu settings.

Unrestricted: all options on LCD menu can be accessed;  
 Basic settings only: option for configuration cannot be displayed;  
 Constraint Mode: options for configuration, factory functions and network cannot be displayed;  
 Locked Mode: MENU is disabled

[Reset to Default](#)

[Undo](#)

- **Functionality**

Add another configuration via keypad menu mode called "Locked Mode". When this option is selected, it will restrict user access to any settings on the LCD: menu is disabled, only allow making and receiving phone calls and basic functionality provided by the feature keys (transfer/conference/hold/mute/volume/send/speaker).



## ENABLE DHCP VLAN

- Web Configuration**

This option can be found under device web UI → Network → Advanced Settings

Enable DHCP VLAN

☐ Enabled ☒ Disabled

- Functionality**

Added support DHCP Option 132 & 133 tunneled through DHCP Option 43 on the phone to deploy it on the network. By enabling the DHCP VLAN, the phone will try to read VLAN setting from DHCP option 132 and 133. If both of values in these two options are empty, phone will then read from option 43. The format of the data in Option 43 should be like as follows:

Code	Len	Data item	Code	Len	Data item	Code
T1	n	d1   d2   ...	T2	n	D1   D2   ...	

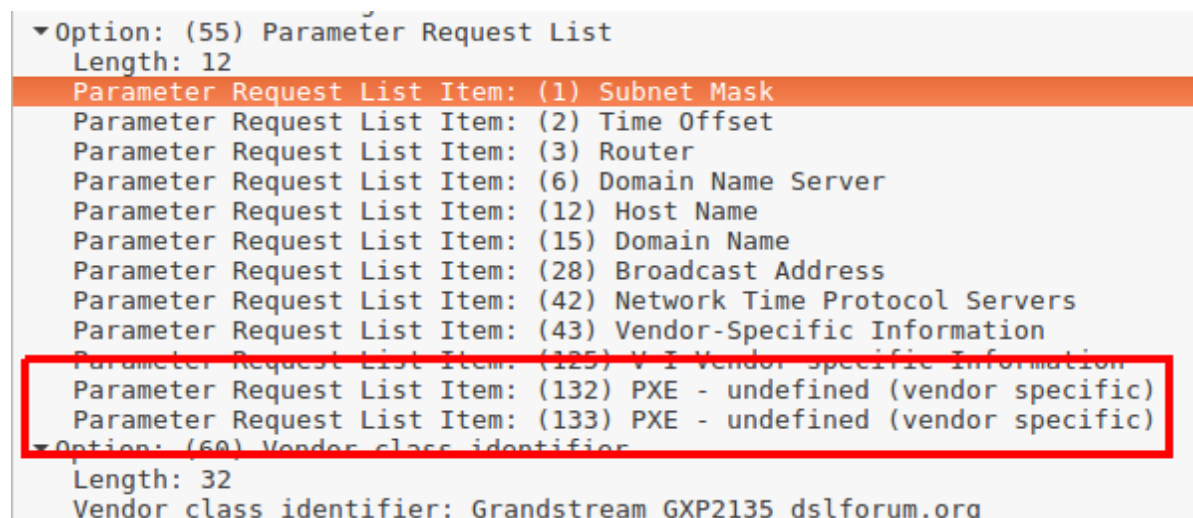
Here is a valid example:

84:04:31:31:34:00:85:02:35:00

Which means tunnel option 132 value 114 and option 133 value 5 via option 43. Option 132 for VLAN ID and 133 for priority, or from Option 43 which encapsulates both two values.

After user enables this feature, phone will disable LLDP VLAN setting and try to read VLAN setting from DHCP to override manual setting.

Firstly, the phone will send discover package and receive offer package for option 132 (VLAN ID) and option 133 (VLAN priority) via DHCP server.



If one of Option 132 and Option 133 is not empty, phone will read values in these two options.

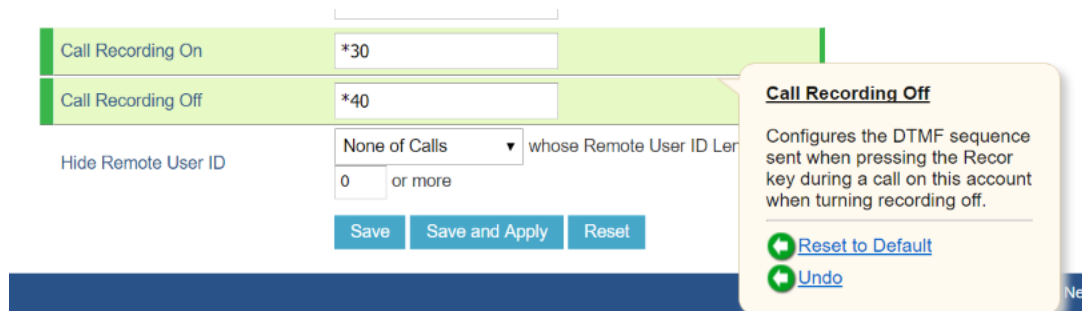
However, if Option 132 and Option 133 are both empty, the phone will retrieve the value of Option 43 for VLAN ID and priority.

If VLAN ID and priority are both valid, VLAN setting will take effect after phone rebooting; if one of the value is invalid, DHCP VLAN setting won't override manual setting.

## ABILITY TO CONFIGURE RECORD BUTTON

- Web Configuration**

This option can be found under webUI →Account→Call Settings.



- Functionality**

Allow users to configure the DTMF that will be sent once record button is pressed during the call.

- New P Value**

Pvalue	Description	Value Range	Default
P26028	<New Pvalue> Call Recording On	String	None
P26029	<New Pvalue> Call Recording Off	String	None

## **FIRMWARE VERSION 1.0.4.17**

### **PRODUCT NAME**

GXP1610, GXP1620, GXP1625, GXP1628 and GXP1630

### **DATE**

09/02/2016

### **ENHANCEMENT**

- Added ability to set forward on busy and on no answer from the LCD
- Added dial soft key on LCD screen
- Added ability to displays more characters for account name on LCD display
- Added Option for Predictive Dial Feature
- Added Option to adjust the Maximum Transmission Unit (MTU) of IP Packets.
- Spited Register and Subscription expiration settings
- Added ability to hide web access mode and disable SSH from LCD menu under constraint mode
- Add support for Latvian Language
- Polish language update

### **BUG FIX**

- Fixed speed dial via active account doesn't work on MPK
- Fixed MPK work abnormally after modifying MPK keys
- Fixed device work abnormally after setting the MPK1
- Fixed phone stops sending keep alive
- Phone Keeps Sending RTP after Call Disconnected
- Fixed cannot park the call with alphanumeric entry
- Fixed Speaker icon displayed when after enabled DND feature
- Fixed cannot get to the voice IP directly after the restart
- Fixed phone ignored assigned VLAN via LLDP
- Fixed device dsp crashed after hold and resume the call on SRTP
- Fixed notify from call group pickup not applied
- Fixed cannot get to the voice IP directly after the restart
- Fixed Contacts disappear after uploading the XML file in not phonebook
- Fixed Local phonebook contacts disappeared

## NEW FEATURES OVERVIEW

This section lists major new features and describes how to use it from the user's point of view.

### SET FORWARD ON BUSY AND ON NO ANSWER FROM LCD

- **Web Configuration**

This setting is under web UI->Settings->LCD Display.

**Settings**  
 General Settings  
 Call Features  
 Multicast Paging  
 Ring Tone  
 Audio Control  
**LCD Display**  
 Date and Time  
 Web Service  
 XML Applications  
 Programmable Keys  
 Broadsoft XSI

## LCD Display

Backlight Brightness: Active

Backlight Brightness: Idle

LCD Contrast

Disable Missed Call Backlight ☒ No ☐ Yes

Wide Idle Screen View ☒ No ☐ Yes

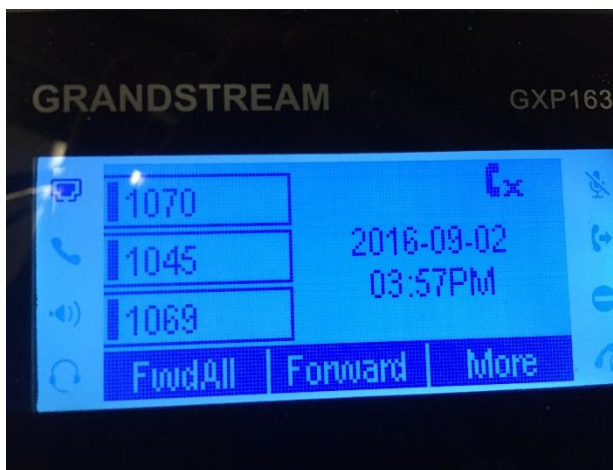
### Softkeys

☐ Missed  
☐ ForwardAll  
☐ Redial  
☐ Forward

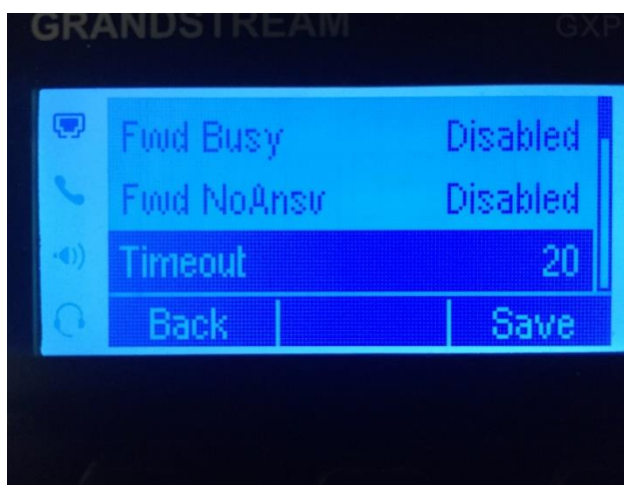
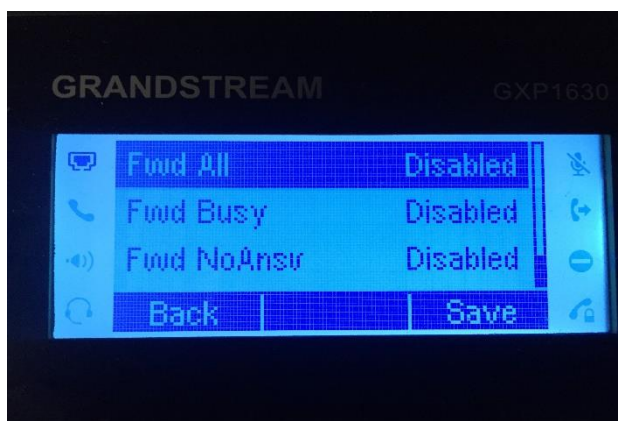
☐ Hide System Softkey on Main Page

- **Functionality**

User can press Forward Softkey on LCD main screen and perform a setting on forward function.



User can select “Fwd All”, “Fwd Busy”, “Fwd NoAnswer” and “Timeout” for Forward for account 1.



- **New P Value**

Pvalue	Description	Value range	Default
P8348	<New Pvalue> Pvalue to set whether to hide System Softkey on Main Page	Checkbox	

### DISPLAY MORE CHARACTERS FOR ACCOUNT NAME ON LCD DISPLAY

- **Web Configuration**

The users could find this setting under web UI->Settings->LCD Display.

### Settings

- General Settings
- Call Features
- Multicast Paging
- Ring Tone
- Audio Control
- LCD Display**
- Date and Time
- Web Service
- XML Applications
- Programmable Keys
- Broadsoft XSI

## LCD Display

---

Backlight Brightness: Active

Backlight Brightness: Idle

LCD Contrast

Disable Missed Call Backlight

☒ No ☐ Yes

Wide Idle Screen View

☒ No ☐ Yes

### Softkeys

Hide System Softkey on Main Page

☐ Missed  
☐ ForwardAll  
☐ Redial  
☐ Forward

Save

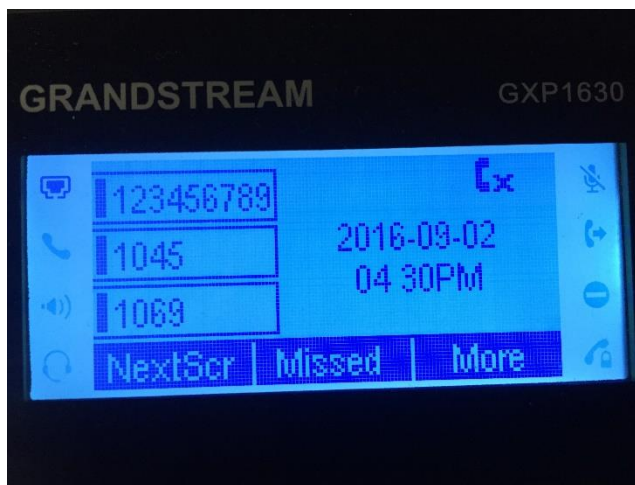
Save and Apply

Reset

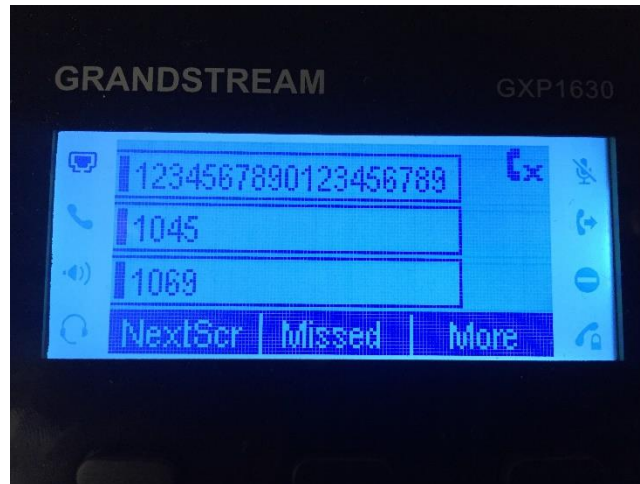
- **Functionality**

User can enable or disable the wide view for account in idle screen. Once this feature is enabled, the LCD could display more characters for account name display. This feature needs a reboot to take effect.

For example, if we set account name for account 1 as "1234567890123456789", before enabling this feature, the phone display looks like this:



After enabling this feature, the phone display looks like this:



- **New P Values**

Pvalue	Description	Value range	Default
P8346	<New Pvalue> Pvalue to set whether to use long label	0, No 1, Yes	0

## ENABLE PREDICTIVE DIAL FEATURE

- **Web Configuration**

User can enable this feature from web UI->Settings->Call Features.

User-Agent Prefix

Predictive Dialing Feature

☐ Disabled
 ☒ Enabled

Save

Save and Apply

Reset

**Predictive Dialing Feature**

Controls the predictive dialing feature on the call screen.

[Reset to Default](#)

[Undo](#)

- **New P Values**

Pvalue	Description	Value range	Default
P22126	<New Pvalue> Pvalue to set whether enable or disable predictive dialing feature	0, Disable 1, Enable	1

## ADJUST EXPIRATION TIME FOR SUBSCRIBE

- **Web Configuration**

This option is under web UI->Account x->SIP Settings->Basic Setting.

**Accounts**

Account 1

General Settings

Network Settings

SIP Settings

**Basic Settings**

Advanced Features

Session Timer

Security Settings

Audio Settings

Call Settings

Account 2

Account 3

## Basic Settings

TEL URI
☒ Disabled
☐ User=phone
☐ Enabled

SIP Registration
☐ No
☒ Yes

Unregister on Reboot
☒ No
☐ All
☐ Instance

Register Expiration

Subscribe Expiration

Reregister before Expiration

Enable OPTIONS Keep Alive
☒ No
☐ Yes

OPTIONS Keep Alive Interval

OPTIONS Keep Alive Max Lost

Local SIP Port

**Subscribe Expiration**

Specifies the frequency (in minutes) in which the phone refreshes its subscription with the specified registrar. The maximum value is 64800 (about 45 days).

[Reset to Default](#)

[Undo](#)

- New P Value**

Pvalue	Description	Value range	Default
P26051	<New Pvalue> Pvalue to set subscribe expiration time for account 1	1-64800	60
P26151	<New Pvalue> Pvalue to set subscribe expiration time for account 2	1-64800	60
P26251	<New Pvalue> Pvalue to set subscribe expiration time for account 3	1-64800	60

## LATVAIN LANGUAGE SUPPORT

- Web Configuration**

This option can be found under device web UI->Maintenance->Language.



**Maintenance**  
 Web Access  
 Upgrade and Provisioning  
 Syslog  
**Language**  
 Action URL  
 TR-069  
 Security

## Language

Display Language
 

English
 

العربية  
 Czech  
 Deutsch  
 English  
 Español  
 Français  
 עברית  
 Hrvatski  
 Magyar  
 Italiano  
 日本語  
 한국어  
**Latviešu valoda**  
 Nederlands  
 Polski  
 Português  
 Русский  
 Slovenščina  
 Svenska  
 Türkçe

Language File Postfix
 

/
 

Reset

### HIDING “WEB ACCESS MODE” AND “DISABLE SSH” FROM LCD MENU UNDER CONSTRAINT MODE

- Web Configuration**

This option can be found under web UI->Maintenance->Security.

**Maintenance**  
 Web Access  
 Upgrade and Provisioning  
 Syslog  
 Language  
 Event Notification  
 TR-069  
**Security**

## Security

Configuration via Keypad Menu
 

☒ Unrestricted
 ☐ Basic settings only
 ☐ Constraint Mode

Enable STAR Key Keypad Locking
 

☒ No
 ☐ Yes

Password to Lock/Unlock

SIP TLS Certificate

In Constraint mode, options for configuration, factory functions and network cannot be display on LCD configuration through keypad.

- Modified P Values**

P value	Description	Value range	Default
P1357	<New P value> P value to set which mode to be configured for Keypad Menu.	0, Unrestricted 1, Basic setting only 2, Constraint Mode	0

## **FIRMWARE VERSION 1.0.4.6**

### **PRODUCT NAME**

GXP1610, GXP1620, GXP1625, GXP1628 and GXP1630

### **DATE**

07/06/2016

### **ENHANCEMENT**

- Added ability to customize the system softkey
- Added ability to hide system softkey on main screen
- Added supports of IEEE-48-addr and IEEE-EUI-64 in PEI header
- Added supports of WAN access node in SIP:PANI header
- Added supports of configurable User Agent field

### **BUG FIX**

- Fixed Broadsoft directory could not display on LCD.
- Fixed GUI module crashed after downloading config file.
- Fixed Download local phonebook should not affect downloading Broadsoft phonebook.
- Fixed Local phonebook/call logs disappear.
- Fixed Mirrored PC port mode works abnormally
- Fixed DUT cannot establish a call after sent SUBSCRIBE REGISTER without response.
- Fixed cannot resume HOLD when pressing HOLD button

### **NEW FEATURES OVERVIEW**

This section lists major new features and describes how to use it from the user's point of view.

#### **CUSTOMIZING SOFTKEY WITH MULTI-PURPOSE FUNCTION ON MAIN SCREEN**

- **Web/LCD Configuration**

Users can configure softkeys from web UI->Settings->Programmable Keys.

**Settings**  
 General Settings  
 Call Features  
 Multicast Paging  
 Ring Tone  
 Audio Control  
 LCD Display  
 Date and Time  
 Web Service  
 XML Applications  
**Programmable Keys**  
 Broadsoft XSI

## Programmable Keys

---

### Line Keys

	Mode	Account	Description	Value
LINE 1	LINE	Account 1	Description	Value
LINE 2	LINE	Account 1	Description	Value

### Softkeys

	Mode	Account	Description	Value
SOFTKEY 1	Default	Account 1	Description	Value
	Speed Dial			
	Speed Dial via Active Account			
	Voice Mail			
	Call Return			
	Intercom			
	LDAP Search			

Save and Apply Reset

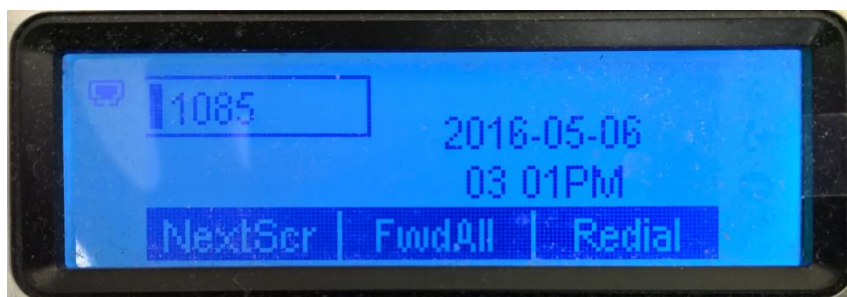
Copyright © Grandstream Networks, Inc. 2016. All Rights Reserved.

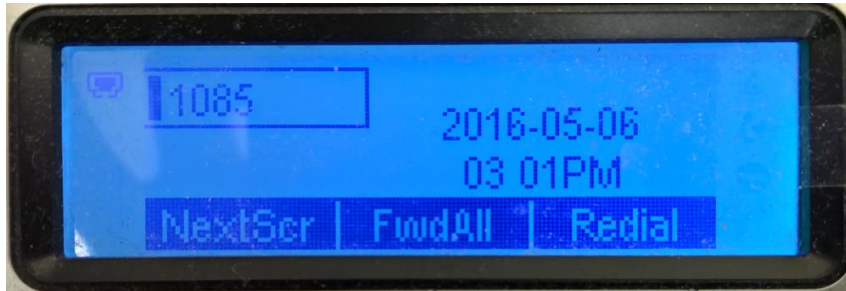
### • Functionality

Added supports for customizing the main screen softkey by assigning a multi-purpose function.

All Grandstream IP Phone softkey design keep the **FIRST** and **LAST** softkeys for special usage. For the first softkey, it is mainly used for main pages switching on idle screen or switching between idle screen and call screen. For the last softkey, it could be used as softkey switching key if the phone couldn't show all available softkeys in one screen. Therefore, available softkey slots for users to customize should not include the first and last softkeys.

For example, the Phone model with 3 softkeys, ex. GXP1630, provide 1 programmable softkey.





#### New P Values

Pvalue	Description	Value range	Default
P2987	Mode	0: Default 10: Speed Dial 14: Speed Dial with Active Account 16: Voice Mail 17: Call Return 20: Intercom 21: LDAP Search	0
P2988	Account	0: SIP account 1 1: SIP account 2 2: SIP account 3	0
P2989	Description	String	None
P2990	Value	String	None

#### HIDE SYSTEM SOFTKEYS ON MAIN SCREEN

- **Web UI/LCD Configuration**

Users can configure softkeys from web UI->Settings->LCD Display.

Settings

General Settings  
 Call Features  
 Multicast Paging  
 Ring Tone  
 Audio Control  

LCD Display

 Date and Time  
 Web Service  
 XML Applications  
 Programmable Keys  
 Broadsoft XSI

## LCD Display

---

Backlight Brightness: Active

6

Backlight Brightness: Idle

2

LCD Contrast

3

Disable Missed Call Backlight

☒ No
 ☐ Yes

## Softkeys

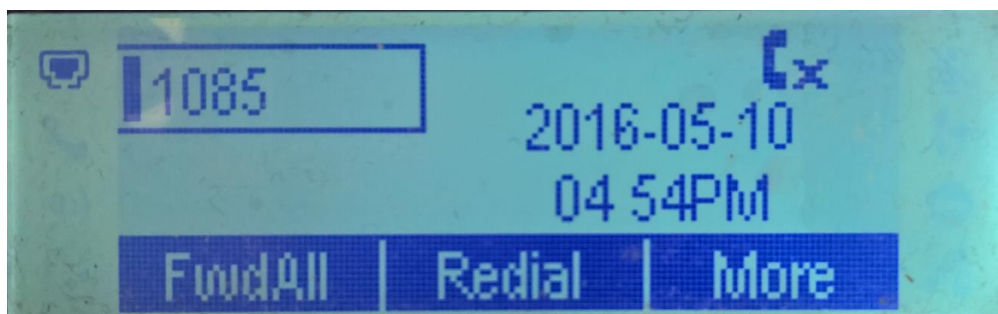
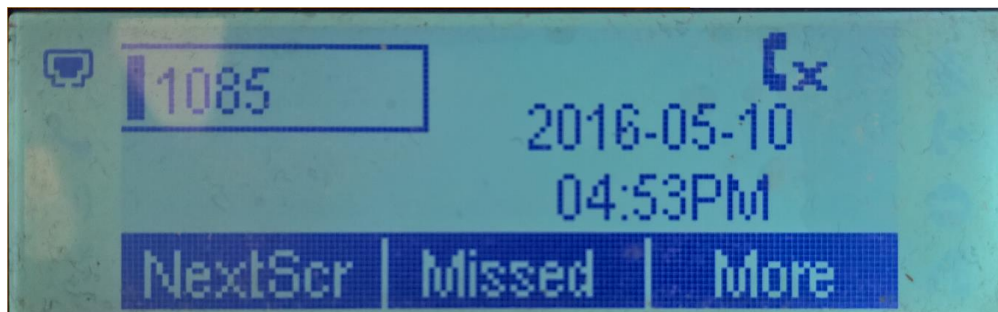
Hide System Softkey on Main Page

☒ Missed  
☒ ForwardAll  
☐ Redial

- **Functionality**

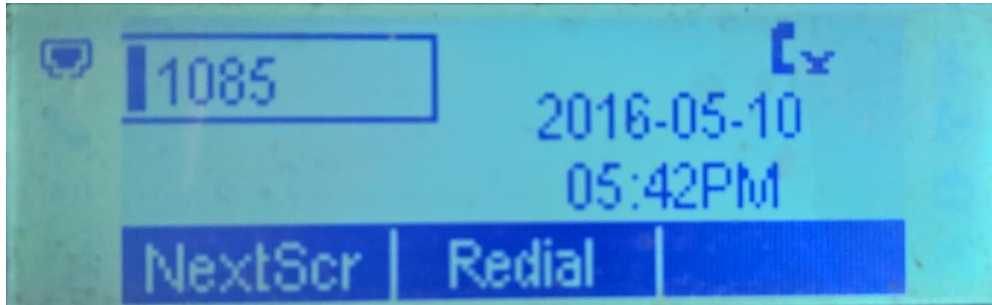
Added support function for user to hide an unnecessary softkeys to make softkey list on the phone less redundant.

Example:



After the configuring the phone to hide Missed and ForwardAll softkeys, phone will only show Redial

softkeyon idle screen.



- **New P Values**

P value	Description	Value range		Default
P8348	Hide System Softkey on Main Page	Option	String	None
		Missed	Missed	
		ForwardAll	ForwardAll	
		Redial	Redial	

## FIRMWARE VERSION 1.0.3.28

### PRODUCT NAME

GXP1610, GXP1620, GXP1625, GXP1628, GXP1630

### DATE

05/20/2016

### ENHANCEMENT

- Added search feature on Broadsoft directories
- Added specific model configuration file
- Added supports for Configure Broadsoft XSI SIP Authentication method by selecting the account

### BUG FIX

- Fixed phone module crashed after registering account and monitoring BLF
- Fixed phone replies with 486 busy here to incoming INVITE during sending subscribe requests
- Fixed LCD screen displays abnormal when dialing an unmatched dialing rule numbers in idle screen
- Fixed the Event Duration in RFC2833 DTMF is unstable which leads failure to check voice mail
- Fixed phone stuck in transferring screen when the blind transfer fails
- Fixed no ring tone in headset when enabling toggle mode and “Always Ring Speaker” option
- Fixed phone cannot hang-up after performing blind transfer in speaker mode
- Fixed Broadsoft n-ways conference call not working
- Fixed “New” contact softkey for local phonebook appeared in Broadsoft phonebook
- Fixed AoC information keep blinking

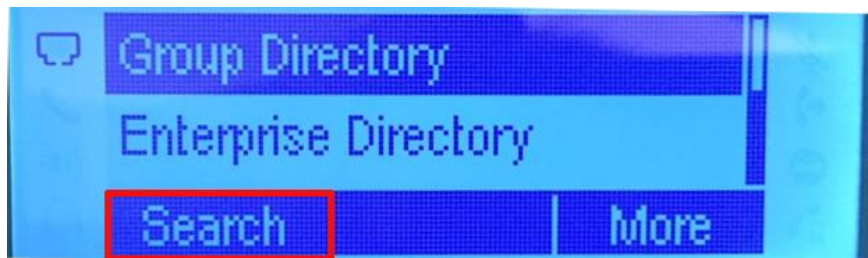
### NEW FEATURES OVERVIEW

This section lists major new features and describes how to use it from the user's point of view.

#### BROADSOFT DIRECTORIES SEARCH FEATURE

- **Functionality**

Added supports for searching Broadsoft XSI directories through phone's LCD. Users can search the name or number of the phonebook.





## SPECIFIC MODEL CONFIGURATION FILE

- Functionality**

Add model specific configuration file to be requested during the provisioning process. For example: When GXP1625 request the provisioning files it need to also request the following file: cfggxp1625.xml.

```

+ Frame 7446: 213 bytes on wire (1704 bits), 213 bytes captured (1704 bits) on interface 0
+ Ethernet II, Src: Grandstr_8f:16:f8 (00:0b:82:8f:16:f8), Dst: All-HSRP-routers_01 (00:00:0c:07:ac:01)
+ Internet Protocol Version 4, Src: 192.168.40.149 (192.168.40.149), Dst: 54.169.67.73 (54.169.67.73)
+ Transmission Control Protocol, Src Port: 54270 (54270), Dst Port: 80 (80), Seq: 1, Ack: 1, Len: 147
+ Hypertext Transfer Protocol
  GET /gs/cfggxp1625.xml HTTP/1.1\r\n
    User-Agent: Grandstream Model HW GXP1625 SW 1.0.3.26 DevId 000b828f16f8\r\n
    Host: fm.grandstream.com\r\n
    Accept: */*\r\n
  \r\n
  [Full request URI: http://fm.grandstream.com/gs/cfggxp1625.xml]
  [HTTP request 1/1]
  [Response in frame: 7453]

```

```

0040  fc 73 47 45 54 20 2f 67 73 2f 63 66 67 67 78 70  .sGET /g s/cfggxp
0050  31 36 32 35 2e 78 6d 6c 20 48 54 54 50 2f 31 2e  1625.xml HTTP/1.
0060  31 0d 0a 55 73 65 72 2d 41 67 65 6e 74 3a 20 47  1..User- Agent: G
0070  72 61 6e 64 73 74 72 65 61 6d 20 4d 6f 64 65 6c  randstre am Model
0080  20 48 57 20 47 58 50 31 36 32 35 20 53 57 20 31  Hw GXP1 625 SW 1
0090  2e 30 2e 33 2e 32 36 20 44 65 76 49 64 20 30 30  .0.3.26 DevId 00
00a0  30 62 38 32 38 66 31 36 66 38 0d 0a 48 6f 73 74  0b828f16 f8..Host
00b0  3a 20 66 6d 2e 67 72 61 6e 64 73 74 72 65 61 6d  : fm.gra ndstream
00c0  2e 63 6f 6d 0d 0a 41 63 63 65 70 74 3a 20 2a 2f  .com..Ac cept: */
00d0  2a 0d 0a 0d 0a  *..

```

## AUTHENTICATION OPTIONS FOR BROADSOFT XSI

- Web/LCD Configuration**

This option is under web UI->Broadsoft->Broadsoft XSI.

### Broadsoft XSI

#### XSI

Server

Port

XSI Authentication Type

#### Login Credentials

Login Username

Login Password

#### SIP Credentials

SIP UserName

SIP User ID

SIP Password

Network Directories

- Functionality**

Provide SIP Authentication option for Broadsoft XSI.

Configure the SIP Credentials and select SIP Credentials as XSI Authentication Type. Phone will offer SIP Credentials when connecting to Broadsoft XSI.

- **New P values**

Pvalue	Description	Value range	Default
P6772	<New Pvalue> Pvalue to set SIP UserName for Broadsoft XSI	String	None
P22034	<New Pvalue> Pvalue to set SIP User ID for Broadsoft XSI	String	None
P6773	<New Pvalue> Pvalue to set SIP Password for Broadsoft XSI	String	None
P22054	<New Pvalue> Pvalue to set XSI Authentication Type for Broadsoft XSI	0 for Login Credentials 1 for SIP Credentials 2 to 4 for SIP Credentials of account 1 to 3.	0

### CONFIGURE BROADSOFT XSI SIP AUTHENTICATION METHOD BY SELECTING THE ACCOUNT

- **Web Configuration**

User can use Authenticate ID and Authenticate Password as SIP Username and SIP Password respectively for SIP account. This feature can be found in web UI->Accounts->Account #->General Settings.

**Accounts**  
 Account 1  
**General Settings**  
 Network Settings  
 SIP Settings  
 Audio Settings  
 Call Settings  
 Feature Codes  
 Account 2  
 Account 3  
 Account 4

## General Settings

Account Active ☐ No ☒ Yes

Account Name

SIP Server

Secondary SIP Server

Outbound Proxy

Backup Outbound Proxy

BLF Server

SIP User ID

**Authenticate ID**

**Authenticate Password**

Authenticate ID and Authenticate Password must be configured if a user selects that account for accessing Broadsoft XSI Directory and Call Log. In Settings Tab -> Broadsoft -> Broadsoft XSI Settings, users need to configure Server Path, Port, account number, SIP User ID and enable the XSI directories they would like to access.

**Settings**

General Settings

Call Features

Multicast Paging

Ring Tone

Audio Control

LCD Display

LED Control

Date and Time

Web Service

XML Applications

Programmable Keys +

Extension Boards +

Broadsoft -

**Broadsoft XSI**

Broadsoft IM&P

Outbound Notification +

## Broadsoft XSI

### XSI

Server	xsp1.iop1.broadworks.net
Port	80

XSI Authentication Type

Account 1 ▼  
 Login Credentials  
 SIP Credentials  
 Account 1  
 Account 2  
 Account 3  
 Account 4

**Login Credentials**

Login Username

Login Password

**SIP Credentials**

SIP UserName

SIP User ID

SIP Password

## Network Directories

Type	Name	
Group Directory	Enabled ▼	
Enterprise Directory	Enabled ▼	
Group Common	Enabled ▼	
Enterprise Common	Enabled ▼	
Personal Directory	Enabled ▼	

- Functionality**

This new method for Broadsoft XSI authentication is selecting SIP account instead of manually input SIP Username and SIP Password. By this way, users can save time on configuration. Phone can retrieve Authenticate ID and Authenticate password information from the selected account and use them as SIP

credentials for accessing Broadsoft XSI Directory and Call Log information. Check functionality under Menu Screen: when all required information are set, Broadsoft Phonebook option will appear in Contacts menu, Broadsoft Call Log option will appear in Call History tab.

- **Changed P values**

P Value	Description	Value Range	Default
P22054	<Changed P value> P value to set XSI Authentication Type for Broadsoft XSI	2 to 4 for SIP Credentials of account 1 to 3.	0

## **FIRMWARE VERSION 1.0.3.14**

### **PRODUCT NAME**

GXP1610, GXP1620, GXP1625, GXP1628, GXP1630

### **DATE**

04/21/2016

### **ENHANCEMENT**

- Added ability to configure the Broadsoft XSI SIP Authentication method by selecting the account
- Added BLF call pickup with Barge-In option
- Added ability to upload configuration file through web GUI
- Added predictive dialing feature
- Added contact XML exporter
- Added Spanish input and T9 conversion for Spanish and Russian
- Added ability to only hide the incoming User ID option in web page
- Added ActionURL feature support on GXP1610
- Added ability to prevent the phone from sending DNS then HTTPS requests to ipvideotalk media service
- Added support for iLBC and G723.1 vocoder
- Added ability for both headset and handset model can be heard voice in a calling
- Added callee user id hiding option in web page
- Added DHCP Option 125 for TR-111
- Added ability to show agent status of Broadsoft call center on idle screen
- Added support for Broadsoft XSI authentication type
- Updated new Grandstream web theme
- Added the Broadsoft AoC information can be displayed without blinking on the phone's LCD
- Added DHCP option 150 support

### **BUG FIX**

- Fixed GXP16XX : BLF is not updated correctly
- Fixed Local group name displayed abnormally when contains special character @.
- Fixed [ITSP Voice2Net] NA-E: Phone Locks Up due to SUBSCRIBE sent to IP instead of domain
- Attempt Fixed Gui module crash in dialing page
- Fixed GUI crashed after uploading wrong phonebook file
- Fixed Device could hear loopback audio after far end holding talk
- Fixed Provision module crashed when Firmware Upgrade Path as UCM detect address
- Fixed Gui module crash in dialing page
- Fixed Open LLDP on version 0.12.4.46, Upgrade failed under PingAn's network

- Fixed Phone crash when configuring MUTE key in Broadsoft server
- Fixed BLF key press has no effect during active call
- Fixed Plantronics EHS compatibility issue with GXP1628
- Fixed DUT sends SIP instance ID when set to No, unless if DUT is rebooted
- Fixed GUI crashed after pressing softkey via HTTP API.
- Fixed Back softkey in local group is invalid.
- Fixed LCD should add iLBC and G723 codec
- Fixed Account bar display abnormally when configure line key as MPK.
- Fixed Redial/Call return worked abnormally.
- Fixed Call history out of range.
- Fixed Contacts disappear after hard rebooting
- Fixed device upgrade failed and stopped at "Recovery Incomplete" page.
- Fixed cannot use dial softkey to dial from Broadsoft phonebook
- Fixed Broadsoft XSI search only exact match
- Fixed GXP1630 LCD display abnormally after participant end call in conference.
- Fixed phone responds with SIP 180 when INVITE contains REPLACES header
- Fixed LCD display abnormally after receive 488 then having coming call
- Fixed 1600C\_GUI crash when enter into phone option from LCD
- Fixed EHS Headset configuration was not used on the new menu
- Fixed Phone can't receive incoming calls
- Fixed GXP1600C GUI crash after on-hook
- Fixed phone module crash after standby 3 days with configuring 7 BLF MPKs.
- Fixed device display abnormal when press multicast paging MPK in the dial screen
- Fixed GXP1630 LCD display abnormally in phonebook
- Fixed LCD warning window display abnormally when establish 3WC with G729
- Fixed GXP1630 transfer on conference hang-up is valid in 4WC.
- Fixed GXP1630 device didn't clear previous digits when press line key back to conference
- Fixed LCD didn't popped up reboot window after modify network configuration
- Fixed Press EHS key can't send out when dialing in idle status.
- Fixed phone plays DTMF tone first then starts to record.
- Fixed LCD didn't display number in ringing page
- Fixed Back softkey in voice mail page worked abnormally
- Fixed the hidden remote user id appear after the remote hold the call
- Fixed device didn't exit conference after far end reject call
- Fixed speed dial from "Audio Loopback" page would result in sound lost
- Fixed LCD display abnormally after dial unmatched dial rule numbers
- Fixed LDAP search detail function works abnormal
- Fixed The "Edit and Dial" and "Save to Phonebook" function work abnormal
- Fixed Mute Feature work abnormally in dialing page
- Fixed LCD display null in Broadsoft phonebook
- Fixed LCD skipping issue after configure DHCP Option43/66 Override FW option

- Fixed GXP1625 when both accounts have a line assigned and a conference call is in progress if the user presses the Voice Message hard key then Soft Key back device will freeze.
- Fixed DUT crash when adjusting volume and brightness on LCD
- Fixed Back softkey and Clear All function in Messages don't work
- Fixed GXP1630 device dialing two calls at the same time in conference
- Fixed Call history display abnormally
- Fixed LCD jump error when press Back softkey on the UCM Detect page
- Fixed Gui go to the wrong page at the last element of LDAP search result
- Fixed Device crash when check LDAP return search result
- Fixed only encrypt signal but not RTP which under the CBCOM model.
- Fixed softkey is abnormal on LCD after pressing blank softkey when enabling Auto-Attended Transfer
- Fixed GXP1600C GUI crash when add new entry in phonebook.
- Fixed GXP1630 device work abnormally after far end on hook in conference page
- Fixed GXP1630 audio of Account 3 display abnormally on web page
- Fixed GXP1630 softkey display wrong in dialing page
- Fixed GXP16XX DTMF not recognized by Cisco Gateway.
- Fixed GXP1600C there is no Back softkey in LCD
- Fixed no account direct IP call display issue
- Fixed LCD contrast option is invalid in web page.
- Fixed device upgrade failed from build 0.9.21.45 to 1.0.0.34.
- Fixed LCD display unfriendly after disabling account1.
- Fixed DUT not provisioned using HTTPS server with gs-ca.crt
- Fixed LCD didn't display coming call number in talk.
- Fixed device played loopback audio on idle screen
- Fixed GUI crash after pressing arrow keys and phonebook key in talk.
- Fixed GUI crash after dial timeout
- Fixed GUI crash when check phonebook contact entry
- Fixed Dial softkey in phonebook is invalid.
- Fixed LCD didn't display Predictive Dialing number when dialing in idle status.
- Fixed MPK should be invalid when account didn't registered.
- Fixed DUT could not factory reset if do it as soon as it boot up
- Fixed "Enable OPTIONS Keep Alive" function is abnormal
- Fixed LCD didn't displayed instant message content
- Fixed Device should remove "EndCall" softkey in ringing page
- Fixed softkeys display abnormally after delete phonebook entry.
- Fixed softkeys in phonebook is invalid
- Fixed GXP1630 idle/call screen is different than other models
- Fixed LCD display abnormally after recovery conference.
- Fixed LCD should have text prompt when over voltage.
- Fixed device work abnormally when using wrong password to login web page.
- Fixed LCD display abnormally when phonebook contains Russian characters.

- Fixed Click-To-Dial icon display wrong in web page.
- Fixed uploading phonebook file from web page is invalid
- Fixed web page display abnormally when phonebook group contains special characters.
- Fixed LCD should giving a warning window in LDAP when searching null value.
- Fixed LCD didn't have EndCall softkey in talk.
- Fixed device download screen file failed when configure path with protocol head.
- Fixed device should limit the number of groups from web page.
- Fixed switch language is invalid.
- Fixed device can't add new group from LCD after deleting all groups.
- Fixed MPK should be invalid when account didn't registered.
- Fixed down arrow key doesn't work after pressing it in conference interface
- Fixed BLF keys not showing correct status for monitored extension with multiple registration
- Fixed LCD displays name wrong when phonebook number contains "(" and "-" character
- Fixed no account displays on LCD when it is dialing
- Fixed contact displays abnormally on LCD when number is too long
- Fixed Holdcall command is invalid via HTTP API.
- Fixed GUI crashed after pressing up arrow key in hold page.
- Fixed stutter tone didn't work
- Fixed BkSp key doesn't work when filling in one number in dialing interface
- Fixed some options don't display on Audio Settings web page of Account 3
- Fixed phone didn't display account status in web page.
- Fixed phone didn't have default value for core dump option.
- Fixed web page display abnormally when language with Russian.
- Fixed phonebook is abnormal on web page
- Fixed web display extra content in Programmable Keys page.
- Fixed gxp16xx - core download behavior is inconsistent
- Fixed LCD display abnormally in transfer page.
- Fixed GXP1600C\_Web page displayed Preferred Vocoder - choice options wrong.
- Fixed there is no LDAP softkey in dialing page
- Fixed devices upgrade in cycle
- Fixed Device freeze when delete instant messages.
- Fixed GXP1630 line 3 key doesn't work
- Fixed MPK work abnormally when setting it as Busy Lamp Field
- Fixed LEDs didn't lights up in Diagnostic Mode.
- Fixed device failed to adjust the volume
- Fixed Upload phonebook file from web page is invalid. Add support for app side export
- Fixed LCD didn't display favorite contact after clearing digits in dial page
- Fixed Device could not dial favorite contact in page mode
- Fixed Setup completed option in Event Notification worked abnormally
- Fixed GXP1625 not honoring DHCP option 66



- Fixed DUT can't enter dialing interface after pressing EHS headset when headset type is "Plantronics EHS"
- Fixed device upgrade failed, stopped at "Recovery Incomplete" page
- Fixed programmable keys can't be set on web page

## NEW FEATURES OVERVIEW

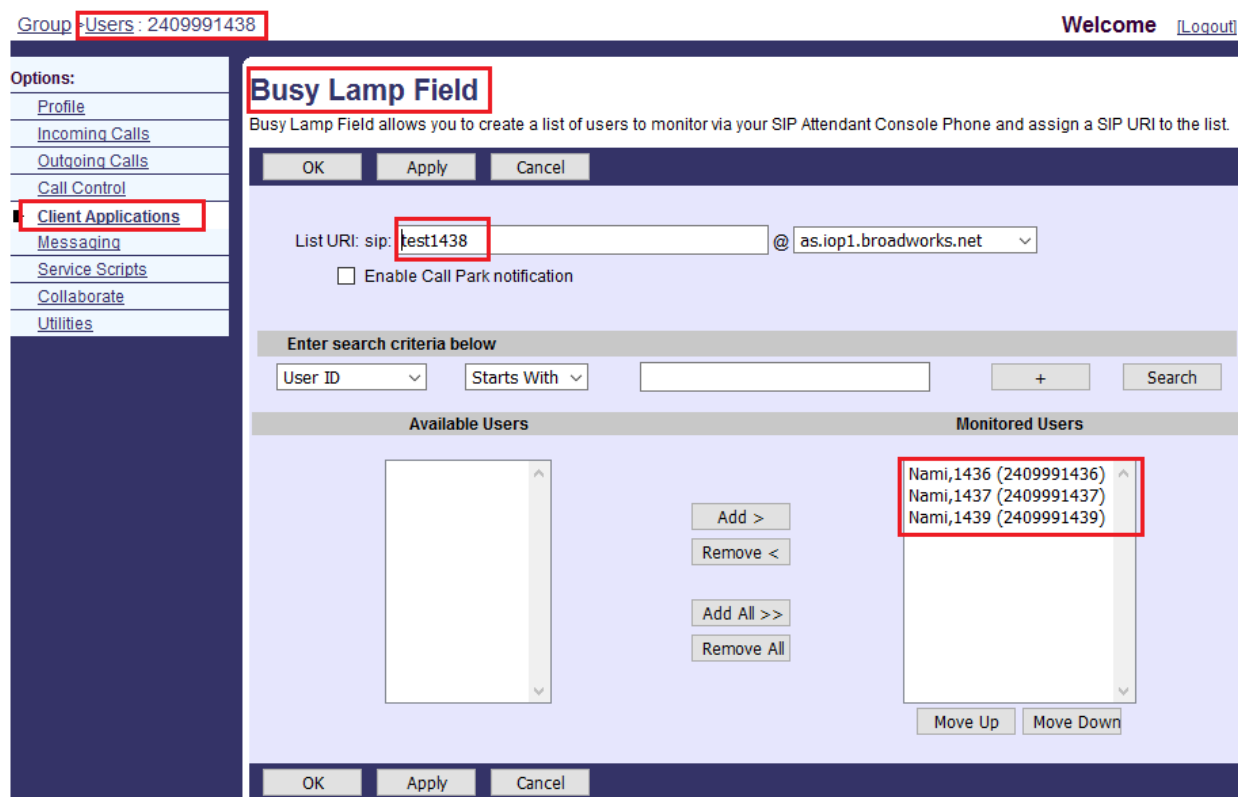
This section lists major new features and describes how to use it from the user's point of view.

### BLF CALL PICKUP BARGE-IN OPTION

- **Web Configuration**

Take Broadsoft service server configuration for example:

Enable BLF Eventlist function from Broadsoft server web UI->Group->Users->Client Applications->Busy Lamp Field.



The screenshot shows the 'Busy Lamp Field' configuration page in the Grandstream web UI. The breadcrumb trail at the top indicates the path: Group -> Users: 2409991438. The left sidebar contains a menu with 'Client Applications' highlighted. The main content area is titled 'Busy Lamp Field' and includes a description: 'Busy Lamp Field allows you to create a list of users to monitor via your SIP Attendant Console Phone and assign a SIP URI to the list.' Below this, there are 'OK', 'Apply', and 'Cancel' buttons. The 'List URI' field is set to 'sip: test1438 @ as.iop1.broadworks.net'. There is an unchecked checkbox for 'Enable Call Park notification'. A search section titled 'Enter search criteria below' includes dropdowns for 'User ID' and 'Starts With', followed by a search button. The interface is divided into two columns: 'Available Users' (empty) and 'Monitored Users' (containing three entries: 'Nami,1436 (2409991436)', 'Nami,1437 (2409991437)', and 'Nami,1439 (2409991439)'). Between the columns are buttons for 'Add >', 'Remove <', 'Add All >>', and 'Remove All'. Below the 'Monitored Users' list are 'Move Up' and 'Move Down' buttons. At the bottom, there are 'OK', 'Apply', and 'Cancel' buttons.

The information List URI sip and Monitored Users field will be configured in Grandstream phone. Check Call Pickup with Barge-in code from Broadsoft server->Group->User->Utilities->Feature Access Codes.

Group: **Users : 2409991438** Welcome [Logout](#)

**Options:**

- [Profile](#)
- [Incoming Calls](#)
- [Outgoing Calls](#)
- [Call Control](#)
- [Client Applications](#)
- [Messaging](#)
- [Service Scripts](#)
- [Collaborate](#)
- [Utilities](#)

## Feature Access Codes

Feature Access Codes list the star codes for services that you have. To activate a service, hit the \* key and the number followed by the # key. Some require additional information such as a phone number, but you are prompted for that information. You cannot change your feature access codes.

OK
Apply
Cancel

Enable	Feature Access Code Name	Main (Required)	Alternate (Optional)
<input checked="" type="checkbox"/>	Advice Of Charge Activation	*34	*A*A*
<input checked="" type="checkbox"/>	Anonymous Call Rejection Activation	*77	
<input checked="" type="checkbox"/>	Anonymous Call Rejection Deactivation	*87	
<input checked="" type="checkbox"/>	Anonymous Call Rejection Interrogation	*52*	
<input checked="" type="checkbox"/>	BroadWorks Anywhere E.164 Dialing	*14	
<input checked="" type="checkbox"/>	Call Bridge	*15	
<input checked="" type="checkbox"/>	Cancel Call Waiting	*70	
<input checked="" type="checkbox"/>	Clear Voice Message Waiting Indicator	*99	
<input checked="" type="checkbox"/>	Connected Line Identification Restriction Interrogation	*56*	
<input checked="" type="checkbox"/>	Customer Originated Trace	*57	
<input checked="" type="checkbox"/>	Directed Call Pickup	*97	
<input checked="" type="checkbox"/>	Directed Call Pickup with Barge-in	*33	
<input checked="" type="checkbox"/>	Direct Voice Mail Transfer	*55	

Based on Broadsoft Service server, the default Feature Access Code for Directed Call Pickup With Barge-in is \*33. This information will be configured in Web Setting-> Accounts -> Account # -> SIP Settings -> Advanced Features.

### Accounts

- Account 1
- General Settings
- Network Settings
- SIP Settings
- Basic Settings
- Advanced Features
- Session Timer
- Security Settings
- Audio Settings
- Call Settings
- Feature Codes
- Account 2

## Advanced Features

Line-seize Timeout
15

Eventlist BLF URI
test1438

Auto Provision Eventlist BLFs
☒ Disabled ☐ Enabled

Conference URI

Music On Hold URI

Force BLF Call-pickup by prefix
☒ Disabled ☐ Enabled

BLF Call-pickup Prefix
\*\*

Call Pickup Barge-In Code
\*33

Configure Eventlist BLF for one of the monitored extension in Multi-Purpose Keys.

## Programmable Keys

### Line Keys

	Mode	Account	Description	Value
LINE 1	LINE ▼	Account 1 ▼	Description	Value
LINE 2	LINE ▼	Account 1 ▼	Description	Value

### Multi-Purpose Keys

	Mode	Account	Description	Value
MPK 1	Eventlist BLF ▼	Account 1 ▼	Description	2409991439
MPK 2	Speed Dial ▼	Account 1 ▼	Description	Value
MPK 3	Speed Dial ▼	Account 1 ▼	Description	Value
MPK 4	Speed Dial ▼	Account 1 ▼	Description	Value
MPK 5	Speed Dial ▼	Account 1 ▼	Description	Value
MPK 6	Speed Dial ▼	Account 1 ▼	Description	Value
MPK 7	Speed Dial ▼	Account 1 ▼	Description	Value
MPK 8	Speed Dial ▼	Account 1 ▼	Description	Value

- Functionality**

This new feature will provide the following functionality:

Allow users to dial a feature access code (FAC) followed by an extension number to pick up or barge-in a call to the monitored extension. If the monitored extension is ringing state (LED indicator is flashing red), a user can pick up that call before the call is answered by monitored side.

If the monitored extension is in talk (LED indicator is solid red), a user can barge-in that existing call. That is, monitored side will release the call, the barge-in user become the controller of this call. When a barge-in occurs and the user's warning tone is enabled on server side, a user would hear barge-in warning tone.

Monitored Extension 2409991429 is Call Ringing State, MPK's LED will be red blinking or monitored extension 2409991429 LED is solid red, which is on talk state. In the above two states, User can use Call Pickup with Barge-in by pressing the MPK, which is configured as Eventlist BLF function key.

When the phone performs Call Pickup with Barge-in, the called number be added \*33 Call Pickup Barge-in code and the user might hear the warning tone if the function is enabled.

## UPLOAD CONFIGURATION FILE THROUGH WEB INTERFACE

- **Web Configuration**

User can find this feature on web UI->Maintenance->Upgrade and Provisioning->Config.



- **Functionality**

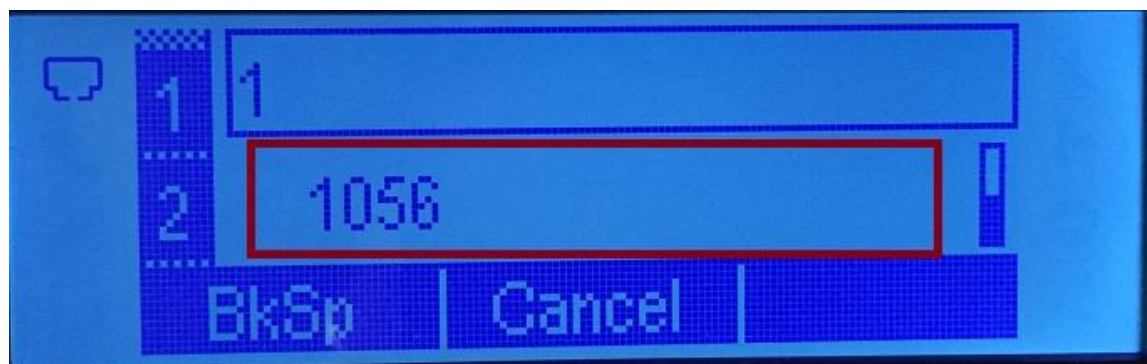
User can upload the config.txt file which can be downloaded from the “Download Device Configuration” link.

Note: the uploaded file must be a text format file.

## PREDICTIVE DIALING FEATURE

- **Functionality**

User can will see the predictive number that might be matched with the number which has been pressed. The numbers in call history and phone book will be matched. User can press up and down Key to see all the matched list.



## EXPORT CONTACT XML

- **Web Configuration**

User can find this feature on web UI->Phonebook->Phonebook->Management->Download XML Phonebook link.

## Phonebook Management

Enable Phonebook XML Download	<input type="text" value="Disabled"/>
Phonebook XML Server Path	<input type="text"/>
Phonebook Download Interval	<input type="text" value="0"/>
Remove Manually-edited Entries on Download	<input type="radio"/> No <input checked="" type="radio"/> Yes
Sort Phonebook by	<input checked="" type="radio"/> Last Name <input type="radio"/> First Name
Download XML Phonebook	<a href="#">Download</a>
Upload XML Phonebook	<a href="#">Upload</a>
Phonebook Key Function	<input type="text" value="Default"/>
<a href="#">Save</a> <a href="#">Save and Apply</a> <a href="#">Reset</a>	

- Functionality**

User can use Download XML Phonebook link to download the current stored phonebook, a phonebook.xml file will be downloaded to local computer.

## HIDE REMOTE USER ID VIA WEB PAGE

- Web Configuration**

User can find this feature on web UI->Accounts->Account #->Call Settings.

DND Call Feature On	<input type="text"/>
DND Call Feature Off	<input type="text"/>
Hide Remote User ID	<input type="text" value="Both Direction Calls"/> whose Remote User ID Length is <div> <input checked="" type="radio"/> Both Direction Calls         <input type="radio"/> None of Calls         <input type="radio"/> Incoming Calls         <input type="radio"/> Outgoing Calls         <input type="radio"/> Both Direction Calls       </div>
<a href="#">Reset</a>	

**Hide Remote User ID**

Allows users to hide the number and display name of remote user ID more than the specific length in an incoming/outgoing call.

[Reset to Default](#)

[Undo](#)

- Functionality**

Allows users to hide the number and display name of remote user ID more than the specific length in an incoming/outgoing call. LCD should display user id as \*\*\*\* when remote user id matched the condition.

## ACTION URL FEATURE SUPPORT ON GXP1610

- Web Configuration**

User can find this new feature in web UI->Maintenance->Event Notification.

## Event Notification

---

Setup Completed	<input type="text"/>
Registered	<input type="text"/>
Unregistered	<input type="text"/>
Off Hook	<input type="text"/>
On Hook	<input type="text"/>
Incoming Call	<input type="text"/>
Outgoing Call	<input type="text"/>
Missed Call	<input type="text"/>
Established Call	<input type="text"/>
Terminated Call	<input type="text"/>
Open DND	<input type="text"/>
Close DND	<input type="text"/>
Open Forward	<input type="text"/>
Close Forward	<input type="text"/>
Blind Transfer	<input type="text"/>
Attended Transfer	<input type="text"/>
Hold Call	<input type="text"/>
UnHold Call	<input type="text"/>
Open Syslog	<input type="text"/>
Close Syslog	<input type="text"/>

[Save](#)[Save and Apply](#)[Reset](#)

- **Functionality**

Send the Action URL when the IP phone finishes the assigned action.

## ABILITY TO PREVENT THE PHONE FROM SENDING DNS THEN HTTPS REQUESTS TO IPVIDEOTALK MEDIA SERVICE

- **Web Configuration**

User can find this new feature in Settings->Web service->Enable IPVideoTalk.

- **Functionality**

User can enable or disable Grandstream IPVideoTalk services on the phone. Reboot is required. After disabling this feature, the phone doesn't send DNS queries to IPVideoTalk media service.

- **New P values**

Pvalue	Description	Value range	Default
P8338	<New P-value> P-value to enable or disable Grandstream IPVdeioTalk services on the phone.	0, No 1, Yes	1

## SUPPORT ILBC AND G723.1 VOCODER

- **Web Configuration**

User can find this new feature in web UI->Accounts->Account Number->Audio Settings.

### Audio Settings

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

DTMF Payload Type

Preferred Vocoder - choice 1

Preferred Vocoder - choice 2

Preferred Vocoder - choice 3

Preferred Vocoder - choice 4

Preferred Vocoder - choice 5

Preferred Vocoder - choice 6

Preferred Vocoder - choice 7

- **Functionality**

Configure vocoders iLBC and G723 in a preference list that is included with the same preference order in SDP message.

- **New/Changed P values**

Pvalue	Description	Value range	Default
P57 P451 P551	<New Pvalue range > Pvalue to set Preferred Vocoder – choice 1	4: G.723.1 <New added> 18: G.729A/B 0: PCMU 8: PCMA 2: G.726-32 9: G.722(wide band) 98: iLBC <New added>	0
P58 P452 P552	<New Pvalue range > Pvalue to set Preferred Vocoder – choice 2	4: G.723.1 <New added> 18: G.729A/B 0: PCMU 8: PCMA 2: G.726-32 9: G.722(wide band) 98: iLBC <New added>	8
P59 P453 P553	<New Pvalue range /Default Pvalue changed> Pvalue to set Preferred Vocoder – choice 3	4: G.723.1 <New added> 18: G.729A/B 0: PCMU 8: PCMA 2: G.726-32 9: G.722(wide band) 98: iLBC <New added>	Changed to 4
P60 P454 P554	<New Pvalue range> Pvalue to set Preferred Vocoder – choice 4	4: G.723.1 <New added> 18: G.729A/B 0: PCMU 8: PCMA 2: G.726-32 9: G.722(wide band) 98: iLBC <New added>	18



P61 P455 P555	<New Pvalue range> Pvalue to set Preferred Vocoder – choice 5	4: G.723.1 <New added> 18: G.729A/B 0: PCMU 8: PCMA 2: G.726-32 9: G.722(wide band) 98: iLBC <New added>	9
P62 P456 P556	<New Pvalue added> Pvalue to set Preferred Vocoder – choice 6	4: G.723.1 18: G.729A/B 0: PCMU 8: PCMA 2: G.726-32 9: G.722(wide band) 98: iLBC	98
P63 P457 P557	<New Pvalue added> Pvalue to set Preferred Vocoder – choice 7	4: G.723.1 18: G.729A/B 0: PCMU 8: PCMA 2: G.726-32 9: G.722(wide band) 98: iLBC	2

## **FIRMWARE VERSION 1.0.2.27**

### **PRODUCT NAME**

GXP1610, GXP1620, GXP1625, GXP1628, GXP1630

### **DATE**

11/19/2015

### **BUG FIX**

- FixedAdjusted Hebrew Font to a smaller size
- Fixed Hebrew strings update

## **FIRMWARE VERSION 1.0.2.26**

### **PRODUCT NAME**

GXP1610, GXP1620, GXP1625, GXP1628, GXP1630

### **DATE**

10/28/2015

### **ENHANCEMENT**

- Added ability to filter characters from dialed numbers

### **BUG FIX**

- Fixed LCD doesn't have reconf softkey after answer incoming call in conference in certain scenario
- Fixed LCD display abnormal after receiving 488 then having incoming call
- Fixed EndCall softkey is invalid when establishing conference using soft pause on MPK speed dial number
- Fixed There is no ring tone in incoming call when adjusting volume
- Fixed Click-To-Dial first available account feature works abnormally
- Fixed BLF keys not showing correct status for monitored extension with multiple registrations

## **FIRMWARE VERSION 1.0.2.25**

### **PRODUCT NAME**

GXP1610, GXP1620, GXP1625, GXP1628, GXP1630

### **DATE**

10/14/2015

### **ENHANCEMENT**

- Updated logo for web UI

### **BUG FIX**

- Fixed DUT not provisioned using HTTPS server with gs-ca.crt

## **FIRMWARE VERSION 1.0.2.22**

### **PRODUCT NAME**

GXP1610, GXP1620, GXP1625, GXP1628

### **DATE**

09/03/2015

### **ENHANCEMENT**

- Added DHCP Option 125 support

### **BUG FIX**

- Fixed DTMF not recognized by Cisco Gateway
- Fixed Inverted LCD Hebrew Strings in Account Name and Incoming Calls
- Fixed Phone plays DTMF tone first then starts to record
- Fixed LCD warning "File Exists" when no 802.1x CA Certificate file uploaded
- Fixed Pressing EHS key can't send out when dialing in idle status
- Fixed LCD doesn't prompt reboot window after modifying network configuration
- Fixed Device displays abnormal when pressing multicast paging MPK in the dial screen

## **FIRMWARE VERSION 1.0.2.21**

### **PRODUCT NAME**

GXP1610, GXP1620, GXP1625, GXP1628

### **DATE**

08/21/2015

### **ENHANCEMENT**

- Split upgrading method, username and password settings for firmware upgrading and Provisioning
- Improved security by hiding password from command line
- Added option to disable SUBSCRIBE message for x-broadworks-callpark event when it's not supported or enabled for the Broadsoft server

### **BUG FIX**

- Fixed "Back" softkey doesn't work
- Fixed Downloading idle screen file abnormally
- Fixed LCD language should be Chinese after factory reset if device obtains IP address from China location
- Fixed Phone could not hear multicast paging sender if the multicast paging call is answered on audio loopback interface
- Fixed Audio channel is incorrect if switching audio channel on conference number input interface
- Fixed Audio channel is incorrect after answering incoming call
- Fixed Audio channel is incorrect if pressing MUTE key in transfer number input interface
- Fixed Audio channel is incorrect after pressing ReConf key to recover conference
- Fixed Multicast paging call will be saved into call history if making the call in ForwardAll interface
- Fixed Language display abnormal on LCD when using Czech
- Fixed Phone cannot send HTTP request to download idle screen file when it receives NOTIFY

## **FIRMWARE VERSION 1.0.2.4**

### **PRODUCT NAME**

GXP1610, GXP1620, GXP1625, GXP1628

### **DATE**

05/19/2015

### **ENHANCEMENT**

- Added option to bypass dial plan
- Added configuration option to ignore Alert-Info header when used for distinctive ringtone
- Updated French, Spanish and Hebrew language strings

### **BUG FIX**

- Fixed The Event Duration in RFC2833 DTMF is unstable which leads failure to check voice mail
- Fixed Device crashes after enabling SIP OPTIONS and staying in idle for a long time
- Fixed IP display error when phone is plugged into LLDP environment that has VLAN configured
- Fixed New Zealand time zone needs update
- Fixed LCD displays network and icon in upgrading
- Fixed Missed calls display abnormally
- Fixed Phone behaves abnormally if entering audio loopback when enable HEADSET Key Mode
- Fixed Rejecting a call in "Audio Loopback" interface would result in audio lost
- Fixed The volume is wrong if adjusting it under headset mode
- Fixed Wrong display when trying to get and set static IP, netmask, gateway in SSH access.
- Fixed Handset may play ring back tone when off-hook during volume adjustment
- Fixed LCD freezes on "Downloading firmware file" page in upgrade via TFTP
- Fixed LED on EHS headset won't turn off automatically
- Fixed Provider\_lock doesn't apply to protected P values of account 4 – account 6

## **FIRMWARE VERSION 1.0.1.14**

### **PRODUCT NAME**

GXP1610, GXP1620, GXP1625

### **DATE**

05/06/2015

### **ENHANCEMENT**

- Added LCD prompt support for HTTP digest challenge for provisioning request
- Added support of sending SIP options messages to verify connectivity to the SIP server
- Added dual outbound SIP proxy support
- Added Broadsoft support: outbound proxy failover and fallback feature

### **BUG FIX**

- Fixed Phone freezes on conference interface after long time PCMU conference
- Fixed Phone couldn't hear each other talking when the SSRC is 0



## **FIRMWARE VERSION 1.0.1.12**

### **PRODUCT NAME**

GXP1610, GXP1620, GXP1625

### **DATE**

04/14/2015

### **ENHANCEMENT**

- Added support of DHCP OPTION 160 on Metaswitch FW

### **BUG FIX**

- Fixed handset may play ring back tone when off-hook during adjusting volume
- Fixed LCD freezes on "Downloading firmware file" page in upgrade via TFTP
- Fixed New Zealand time zone need update
- Fixed DNS-SRV connection failed if first option is unavailable using TCP
- Fixed phone could not end call after holding a conference