

# 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 webUl→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, it cannot downgrade to firmware version lower than 1.0.4.50.
- Starting from 1.0.3.28 weather service is removed and no longer supported.



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## **FIRMWARE VERSION 1.0.4.88**

### **PRODUCT NAME**

GXP1610, GXP1615, GXP1620, GXP1625, GXP1628 and GXP1630

#### **DATE**

02/02/2018

#### **BUG FIX**

- Updated Grandstream product certificate
- Fixed Device couldn't upgrade to build 1.0.1.52 from 1.0.1.30 by HTTPS
- Fixed Device gets stuck in "Firmware update" page during HTTPS upgrading process
- Fixed Device works abnormal after registering with Freeswitch account without SIP TLS Private Key Password



### FIRMWARE VERSION 1.0.4.82

#### **PRODUCT NAME**

GXP1610, GXP1615, GXP1620, GXP1625, GXP1628 and GXP1630

#### **DATE**

11/20/2017

#### **ENHANCEMENT**

- Added "Account Display" option to configure SIP account display label on LCD. [ACCOUNT DISPLAY CONFIGURATION]
- Added "Disable VM/MSG Power Light Flash" option to enable/disable voicemail/message indication.
   [ENABLE/DISABLE VOICEMAIL INDICATION]
- Added web UI option to upload SSH public key for SSH access. [SUPPORT ACCESS SSH USING AUTHENTICATION KEYS]
- Added "Ring for Call Waiting" option to enable/disable ringing the speaker phone on call waiting.
   [SUPPORT RING FOR CALL WAITING]
- Added call forwarding configuration on web UI. [SUPPORT CALL FORWARD CONFIGURATION ON WEB UI]
- Added support for static/dynamic auto-attended transfer. [SUPPORT STATIC/DYNAMIC AUTO-ATTENDED TRANSFER]
- Added support for BLF server. [SUPPORT BLF SERVER CONFIGURATION]
- Added automatic redial support. [SUPPORT AUTOMATIC REDIAL]
- Added HTTP/HTTPS Username/Password configuration on LCD menu for firmware upgrading and provisioning. [SUPPORT HTTP(S) USERNAME/PASSWORD CONFIGURATION ON LCD FOR UPGRADE/CONFIG]
- Added blacklist/whitelist support in phonebook. [SUPPORT BLACKLIST/WHITELIST IN LOCAL PHONEBOOK]
- Added option to disable hook switch. [SUPPORT ENABLE/DISABLE HOOKSWITCH]
- Added notification tone value configuration [SUPPORT NOTIFICATION TONE VOLUME]
- Added SNMP support [SUPPORT SNMP]
- Added support for Genesys Agent Login/Logout and status update [SUPPORT GENESYS AGENT LOGIN/LOGOUT/STATUS UPDATE]
- Improved slow performance issue after phone is used for some time.
- Added Let's Encrypt Root CA to Trusted CA
- Increased max length of dial plan from 512 to 1024 characters



#### **BUG FIX**

- Fixedphone is unable to generate second dial tone using dialplan
- Fixed local phonebook does not update the group assigned to new contact
- Fixed the standby server registration address displays incorrectly on the phone
- Fixed phone only shows 50 contacts for Broadsoft phonebook
- Fixed onhook dialing window disappears when there is an incoming call
- · Fixed no audio issue when using with Genesys system
- Fixed multicast paging compatibility issue with other devices

#### **NEW FEATURES OVERVIEW**

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

#### ACCOUNT DISPLAY CONFIGURATION

#### Web Configuration

This option can be found under device web UI→Accounts->Account X->General Settings->Account Display.



#### Functionality

The user can configure how the account label is displayed on the phone for the SIP account. If set to "User Name", LCD account label will display the Account Name configured for this SIP account. If set to "User ID", it will then display the SIP User ID configured for this SIP account.

#### New Pvalue

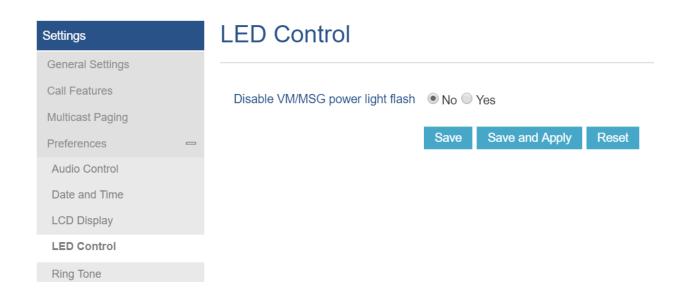
Pvalue	Description	Value Range	Default
P2380 (Account 1) P2480 (Account 2) P2580 (Account 3)	Account Display	Value = 0; User Name Value = 1; User ID	User Name

#### **ENABLE/DISABLE VOICEMAIL INDICATION**

#### Web Configuration

This option "Disable VM/MSG Power Light Flash" can be found under device web UI→Settings->Preference->LED Control.





This option controls the LED behavior upon new/unread Voicemail and Message. If set to "No", the LED light will be flashing when there is unread Voicemail and Message. Otherwise, it will be off.

#### New Pvalue

Pvalue	Description	Value Range	Default
P8371	Disable VM/MSG Power Light Flash	Value = 0; No Value = 1; Yes	No

### SUPPORT ACCESS SSH USING AUTHENTICATION KEYS

#### Web Configuration

The upload option for "SSH Public Key" can be found under device web UI→Maintenance->Security.

SSH Public Key Upload Delete

#### Functionality

This option allows user to use authentication keys for SSH access. The public key should be loaded to phone's web UI while the private key should be used in the SSH tool side. This will allow upcoming SSH access without password.

#### SUPPORT RING FOR CALL WAITING

#### Web Configuration

The option "Ring for Call Waiting" can be found under device web UI→Settings->Call Features.





This option allows user to configure whether the speaker phone will ring when there is new incoming call during an active call. If set to No, the user will hear call waiting tone. If set to Yes, the phone will ring using the speakerphone.

#### New Pvalue

Pvalue	Description	Value Range	Default
P22102	Ring For Call Waiting	Value = 0; No Value = 1; Yes	No

#### SUPPORT CALL FORWARD CONFIGURATION ON WEB UI

### • Web Configuration

Call forwarding configurations can be found under device web UI→Accounts->Account x->Feature Codes.

Accounts	Feature Codes	
Account 1 —		
General Settings	Call Forward Uncondition	onally (All)
Network Settings		
SIP Settings 수	Target	
Audio Settings	Call Forward Busy	
Call Settings		
Feature Codes	Target	
Account 2	Call Forward Delayed (	No Answer)
	Target	
	raiget	
	Delayed Call Forward Wait Time	20

### Functionality

In this web UI page, users can set Call Forward Unconditionally, Call Forward Busy and Call Forward Delayed (No Answer) for each SIP account.



#### New Pvalue

Pvalue	Description	Value Range	Default
forward_0 (Account 1) forward_1 (Account 2) forward_2 (Account 3)	Call Forward Unconditionally (ALL) Target	String	None
busyForward_0 (Account 1) busyForward_1 (Account 2) busyForward_2 (Account 3)	Call Forward Busy Target	String	None
delayedForward_0 (Account 1) delayedForward_1 (Account 2) delayedForward_2 (Account 3)	Call Forward Delayed (No Answer) Target	String	None
P139 (Account 1) P470 (Account 2) P570 (Account 3)	Call Forward Delayed (No Answer) Delayed Call Forward Wait Time	Number	20

#### SUPPORT STATIC/DYNAMIC AUTO-ATTENDED TRANSFER

### Web Configuration

The option is "Attended Transfer Mode" which can be found under device web UI→Settings->Call Features.

Attended Transfer Mode	Static    Dynamic
------------------------	-------------------

### Functionality

When "Attended Transfer Mode" is set to "Static", only "Blind Transfer" softkey is available on the call screen. If set to "Dynamic", both "Blind Transfer" and "Attended Transfer" softkeys are available.

#### New Pvalue

Pvalue	Description	Value Range	Default
P1376	Attended Transfer Mode	Value = 0; Static Value = 1; Dynamic	Static

#### **SUPPORT BLF SERVER CONFIGURATION**

#### Web Configuration

The "BLF Server" optioncan be found under device web UI→Accounts->Account X->General Settings.



Secondary SIP Server		
Outbound Proxy		BLF Server
Backup Outbound Proxy		Optional server used for SUBSCRIBE requests only.
BLF Server		Reset to Default
SIP User ID		Undo

When "BLF Server" is configured, the SUBSCRIBE request will be sent to this address for BLF.# BLF Server

#### New Pvalue

Pvalue	Description	Value Range	Default
P2375 (Account 1) P2475 (Account 2) P2575 (Account 3)	BLF Server	String	None

#### SUPPORT AUTOMATIC REDIAL

#### Web Configuration

Three options are added for this feature under web UI->Settings->Call Features.

- "Enable Automatic Redial": Enable/disable automatic redial.
- "Automatic Redial Times": This configures the number the times the phone will try to redial.
- "Automatic Redial Interval": This configures the interval between each automatic redial.



#### Functionality

When the phone dials a number and no answer, if automatic redial is enabled, the phone will prompt for redial. The redial attempt will be made at the "Automatic Redial Interval" for the number of times defined in "Automatic Redial Times".



### New Pvalue

Pvalue	Description	Value Range	Default
P8353	Enable Automatic Redial	Value = 0; No Value = 1; Yes	No
P8354	Automatic Redial Times	Number 1 to 100	10
P8355	Automatic Redial Interval	Number 1 to 360	20

#### SUPPORT HTTP(S) USERNAME/PASSWORD CONFIGURATION ON LCD FOR UPGRADE/CONFIG

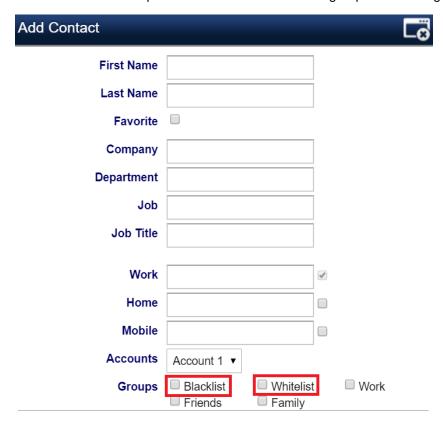
### Functionality

Previously HTTP/HTTPS Username/Password configuration for firmware upgrading/provisioning is only available on phone's web UI. Now this configuration is also available on phone's LCD menu. This setting can be found in LCD menu->System->Upgrade/Config.

### SUPPORT BLACKLIST/WHITELIST IN LOCAL PHONEBOOK

### Web Configuration

Users can add a local phonebook contact to blacklist group or whitelist group.





If the contact number belongs to Blacklist group, the call from this number will be blocked. If the contact number belongs to Whitelist group, when the phone is on DND mode, the call from whitelist number will be allowed.

#### SUPPORT ENABLE/DISABLE HOOKSWITCH

#### Web Configuration

This option "Disable Hook Switch" can be found under web UI->Settings->Call Features.

Disable Hook Switch 

No Yes For Answering Call

### Functionality

When the user disables hook switch, the hook switch will not work to trigger onhook/offhook. If option "Disable Hook Switch" is set to "For Answering Call", the phone's hook switch will not work to answer the call. The user will need to press softkey or speaker/headset/line hardkey to answer the call.

#### New Pvalue

Pvalue	Description	Value Range	Default
P6763	Disable Hook Switch	Value = 0; No Value = 1; Yes Value = 2; For Answering Call	No

#### SUPPORT NOTIFICATION TONE VOLUME

#### Web Configuration

Notification tone volume can be adjusted under web UI->Settings->Ring Tone.



#### Functionality

The volume for paging, intercom, auto answer call and call hold reminder beep notification can be adjusted using this option.

#### New Pvalue

Pvalue	Description	Value Range	Default
P8399	Notification Tone Volume	Value = 0 to 7	4



#### **SUPPORT SNMP**

### Web Configuration

SNMP configuration can be found under phone's web UI->Network->SNMP Settings.

# **SNMP Settings**

Enable SNMP	○ Yes ● No
SNMP Trap Community	
SNMP Trap IP	
SNMP Trap Port	162
SNMP Trap Version	● Version 1 ○ Version 2
SNMP Trap Interval	5

#### New Pvalue

Pvalue	Description	Value Range	Default
P21896	Enable SNMP	Value = 0; No Value = 1; Yes	0
P21900	SNMP Trap Community	String	
P21897	SNMP Trap IP	String	
P21898	SNMP Trap Port	Number: 1025 to 65535	162
P21899	SNMP Trap Version	Value = 0; Version 1 Value = 1; Version 2	Version 1
P21901	SNMP Trap Interval	Number: 1 to 1440	5

### SUPPORT GENESYS AGENT LOGIN/LOGOUT/STATUS UPDATE

#### Web Configuration

The Genesys agent login/logout/status feature can be enabled by below settings. Those options can be found under phone's web UI->Accounts->Account X->SIP Settings->Advanced Features. Notification tone volume can be adjusted under web UI->Settings->Ring Tone.



Enable User Presence Subscription	○ No ● Yes	
Hide Login Soft Key On Auto User Prsence Subscription	● No ○ Yes	
Auto User Presence ID		
Auto User Presence password		

To enable this feature, register Genesys account on GXP16xx's SIP account first. Then on the GXP16xx web UI->Accounts->Account X->SIP Settings->Advanced Features:

- "Enable User Presence Subscription": Yes
- "Hide Login Soft Key On Auto User Presence Subscription": No
- "Auto User Presence ID": The phone will subscribe user presence automatically with this ID once the account is registered.
- "Auto User Presence Password": The phone will subscribe user presence automatically with this password once the account is registered.

A softkey "Genesys" will display on the idle screen. Press the softkey to view and update agent status.

### New Pvalue

Pvalue	Description	Value Range	Default
P2353 (Account 1) P2453 (Account 2) P2553 (Account 3)	Enable User Presence Subscription	Value = 0; No Value = 1; Yes	0
P2354 (Account 1) P2454 (Account 2) P2554 (Account 3)	Hide Login Soft Key On Auto User Presence Subscription	Value = 0; No Value = 1; Yes	1
P2351 (Account 1) P2451 (Account 2) P2551 (Account 3)	Auto User Presence ID	String	
P2352 (Account 1) P2452 (Account 2) P2552 (Account 3)	Auto User Presence Password	String	