GXP1610/1615/1620/1625/1628/1630 Firmware Release Notes

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 webUI→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.

• For GXP16xx with the following hardware versions, once upgraded to firmware 1.0.7.13, downgrading to lower version is not supported:
  GXP1610 is V2.6A or higher,
  GXP1615 is V2.7A or higher,
  GXP1620 is V2.6A or higher,
  GXP1625 is V2.7A or higher,
  GXP1628 is V1.8A or higher,
  GXP1630 is V1.9A or higher.

• Starting from 1.0.3.28 weather service is removed and no longer supported.
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<td>TEST PASSWORD STRENGTH</td>
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<td>NETWORK SECURITY</td>
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<tr>
<td>SUPPORT ACCESS SSH USING AUTHENTICATION KEYS</td>
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<tr>
<td>SUPPORT RING FOR CALL WAITING</td>
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FIRMWARE VERSION 1.0.7.13

PRODUCT NAME
GXP1610, GXP1615, GXP1620, GXP1625, GXP1628 and GXP1630

DATE
2/10/2021

FIRMWARE FILE INFORMATION
• Firmware file name: gxp1600fw.bin
  MD5: 170adb111d83f77bdbe4a5c7e2435259

IMPORTANT UPGRADING NOTE
For GXP16xx with the following hardware versions, once upgraded to firmware 1.0.7.13, downgrading to lower version is not supported:
  GXP1610 is V2.6A or higher,
  GXP1615 is V2.7A or higher,
  GXP1620 is V2.6A or higher,
  GXP1625 is V2.7A or higher,
  GXP1628 is V1.8A or higher,
  GXP1630 is V1.9A or higher.

ENHANCEMENT
• Updated Spanish translation
• Updated German translation
• Updated Italian translation
FIRMWARE VERSION 1.0.7.6

PRODUCT NAME
GXP1610, GXP1615, GXP1620, GXP1625, GXP1628 and GXP1630

DATE
12/04/2020

FIRMWARE FILE INFORMATION
• Firmware file name: gxp1600fw.bin
  MD5: c5f796c60be47ac308ad6b25e9cf9dd8

ENHANCEMENT
• Added Ability to configure variables in the provisioning server URL [SUPPORT OF VARIABLES IN PROVISIONING SERVER URL]

BUG FIX
• Fixed OpenVPN connection does not get established
• Changed Booting message to English when using Hebrew
• Fixed Device does not authenticate correctly with SIP Server
• Fixed GT210 calls fail in secure only mode
• Fixed GT210 calls sometimes do not get terminated

NEW FEATURES OVERVIEW

SUPPORT OF VARIABLES IN PROVISIONING SERVER URL
• Functionality
  The function allows users to configure a unified provisioning server URL for different phone models. Currently the following variables are supported in the provisioning server URL:
    - $PN: it is used to identify the directory name of the provisioning server directory where the corresponding boot files and configuration files are located.
    - $MAC: it is used to identify the MAC address of the IP phone.

  Variables $PN and $MAC can be embedded in server URL setting in Web UI and also in DHCP Option 66.
  Example (Web UI): /192.168.0.2/$PN_$MAC
  Example (DHCP Option 66): tftp://192.168.0.2/$PN_$MAC
  $PN will be replaced with phone model, e.g., GXP1625
  $MAC will be replaced with phone’s MAC address, e.g., 000b829a8ffe
FIRMWARE VERSION 1.0.7.3

PRODUCT NAME
GXP1610, GXP1615, GXP1620, GXP1625, GXP1628 and GXP1630

DATE
09/22/2020

FIRMWARE FILE INFORMATION
• Firmware file name: gxp1600fw.bin
  MD5: c47eb9f461c16bb4a1cd834557a9f29b

ENHANCEMENT
• Added Two accounts support on GXP1610/1615
• Added Ability to auto answer pre-defined numbers [AUTO ANSWER NUMBERS FEATURE]

BUG FIX
• Fixed 802.1x Authentication does not work correctly
• Fixed Device cannot upload OpenVPN client certificate files correctly
• Fixed Device does not contact NTP server from DHCP option 42
• Fixed Device responses to 183 Session Progress message issue
• Fixed Device does not respond to NOTIFY after switching 3CX configuration method
• Fixed Device fails to download 3CX phonebook
• Fixed Authentication information will not be carried when the 3CX account is unregister
• Fixed Device does not update the round trip delay correctly
• Fixed Multicast paging has unstable audio if device is not rebooted
• Fixed Some time zone values are not correct
• Fixed Vulnerabilities CVE-2016-218/QIDS 38657
• Fixed CA certificate not cleared on web GUI
• Fixed Device picking up is slow and no audio when it connects to 2nd server failed using TCP
• Fixed HTTPS provisioning issue via DHCP option 66
• Fixed 3CX auto provisioning issue
• Fixed Account unable to register when the SIP info is input via keypad
• Fixed Device does not send RTP upon receiving “180 Ringing SDP” with a new RTP connection information
• Fixed Device hides Forward softkey after reboot
• Fixed Device does not display 488 response correctly on LCD
NEW FEATURES OVERVIEW

AUTO ANSWER NUMBERS FEATURE

- Web Configuration
  This option can be found under device web UI → Account x → Call Settings.

  Auto Answer
  [ ] No [ ] Yes

  Auto Answer Numbers

- Functionality
  The function allows users to have the phone configured with a pre-defined list of numbers that will perform auto answer.
  There are different situations for auto answer setup:
  1) Auto Answer disable → no auto answer;
  2) Auto Answer enable, no auto answer number specified → all numbers are auto answered;
  3) Auto Answer enable, auto answer number specified → only numbers specified will be auto answered.

  For “Auto Answer Numbers”, it accepts:
  Digits :1,2,3,4,5,6,7,8,9;
  x – any digit from 0-9;
  xx – any two digits from 0-9;
  [1-5] – any digit from 1 to 5;
  +: it matches the previous character as many time as needed like regular expression.
  Please note Auto Answer Numbers can be split with ";", for example: 1x;2xxx;3x+

- Pvalue

<table>
<thead>
<tr>
<th>Pvalue</th>
<th>Description</th>
<th>Value Range</th>
<th>Default</th>
</tr>
</thead>
<tbody>
<tr>
<td>P29064</td>
<td>Auto Answer Numbers</td>
<td>String: Max length is 512.</td>
<td>NULL</td>
</tr>
<tr>
<td>P29164</td>
<td>Auto Answer Numbers</td>
<td>String: Max length is 512.</td>
<td>NULL</td>
</tr>
<tr>
<td>P29264</td>
<td>Auto Answer Numbers</td>
<td>String: Max length is 512.</td>
<td>NULL</td>
</tr>
</tbody>
</table>
FIRMWARE VERSION 1.0.5.3

PRODUCT NAME
GXP1610, GXP1615, GXP1620, GXP1625, GXP1628 and GXP1630

DATE
06/12/2020

FIRMWARE FILE INFORMATION

• Firmware file name: gxp1600fw.bin
  
  MD5: 1c93d58a76aa66eb100c292dad52b324

ENHANCEMENT

• Added Support of Beirut Time zone
• Added Support of Central Africa Time zone “CAT”
• Added Ability to choose TLS Version for HTTPS provisioning [TLS VERSION SELECTION]
• Added Ability to enable/disable weak ciphers [ENABLE/DISABLE WEAK CIPHERS]

BUG FIX

• Fixed BLF indication not working correctly if feature key synchronization is enabled
• Updated the mini_httpd version due to Vulnerabilities CVE-2017-17663
• Fixed Device can’t login web page the first time after switching IP

NEW FEATURES OVERVIEW

TLS VERSION SELECTION

• Web Configuration
This option can be found under device web UI→ Maintenance → Security Settings → Security.

Minimum TLS Version

TLS 1.1

Maximum TLS Version

Unlimited

• Functionality
The function allows users to choose minimum/maximum TLS version for HTTPS provisioning and SIP transport. After change the TLS Version, the phone needs rebooting for taking effect on HTTPS web access. Provisioning and sip transport don’t need reboot.
• **Pvalue**

<table>
<thead>
<tr>
<th>Pvalue</th>
<th>Description</th>
<th>Value Range</th>
<th>Default</th>
</tr>
</thead>
<tbody>
<tr>
<td>P22293</td>
<td>Minimum TLS Version</td>
<td>Value = 10; TLS 1.0&lt;br&gt;Value = 11; TLS 1.1&lt;br&gt;Value = 12; TLS 1.2</td>
<td>TLS 1.1</td>
</tr>
<tr>
<td>P22294</td>
<td>Maximum TLS Version</td>
<td>Value = 99; Unlimited&lt;br&gt;Value = 10; TLS 1.0&lt;br&gt;Value = 11; TLS 1.1&lt;br&gt;Value = 12; TLS 1.2</td>
<td>Unlimited</td>
</tr>
</tbody>
</table>

**ENABLE/DISABLE WEAK CIPHERS**

• **Web Configuration**

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

![Enable/Disable Weak Ciphers](image)

• **Functionality**

This feature could force the TLS version/Cipher suites for HTTPS provisioning and the TLS version for sip transport (TLS/TCP) and HTTPS web access.

• **Pvalue**

<table>
<thead>
<tr>
<th>Pvalue</th>
<th>Description</th>
<th>Value Range</th>
<th>Default</th>
</tr>
</thead>
<tbody>
<tr>
<td>P8536</td>
<td>Enable/Disable Weak Ciphers</td>
<td>Value = 0; Enable Weak TLS Ciphers Suites&lt;br&gt;Value = 1; Disable Symmetric Encryption&lt;br&gt;RC4/DES/3DES&lt;br&gt;Value = 2; Disable Symmetric Encryption SEED&lt;br&gt;Value = 3; Disable All Weak Symmetric Encryption&lt;br&gt;Value = 4; Disable Symmetric Authentication MD5&lt;br&gt;Value = 5; Disable All Weak TLS Cipher Suites</td>
<td>0</td>
</tr>
</tbody>
</table>
FIRMWARE VERSION 1.0.4.152

PRODUCT NAME
GXP1610, GXP1615, GXP1620, GXP1625, GXP1628 and GXP1630

DATE
12/24/2019

FIRMWARE FILE INFORMATION
• Firmware file name: gxp1600fw.bin
  MD5: a28bae43e6d723722152242d2809da

ENHANCEMENT
• Updated Hungarian language display
• Changed "SÃ£o Paulo" time zone to UTC-3
• Added Support to allow more special characters to be used in SIP account via keypad [MORE SPECIAL CHARACTERS VIA KEYPAD]
• Added GNU GPL License information display on LCD and Web UI [GNU GPL LICENSE]
• Added Ability to scroll long names on the screen in order to display the full name [SCROLL LONG NAMES]
• Added Support of “Provision” mode for VPK/MPK to trigger provisioning [PROVISION VPK/MPK]
• Added Support to input username and password via keypad for Broadsoft's directories [BROADSOFT DIRECTORIES]
• Added Support of Chile time zone
• Added Support of an Action URL for screen capture [SCREEN CAPTURE ACTION URI]
• Added Ability to allow @ character in SIP user ID [SIP USER ID FIELD IMPROVEMENT]
• Added Support of Auth Header on Initial REGISTER [ADD AUTH HEADER ON INITIAL REGISTER]
• Add Support to include DHCP Option 12 into DHCP Inform message [HOST NAME(OPTION 12)]
• Increased SIP Certificate length limitation from 2048 characters to 8192

BUG FIX
• Fixed Device freezes on the automatic redial page
• Fixed Devices freezes after uploading malformed phonebook
• Fixed Bypass dial plan is not working properly during API dialing
• Fixed Device does not prompt for authentication when HTTP server has multiple files
• Fixed Device crashes when it answers a call during redial countdown
• Fixed Device freezes on the automatic redial page
• Fixed RecordOn button is displayed as Cancel button on call interface
• Fixed comment error for active backlight timeout
• Fixed Device can’t use phone book dialing when automatic redial prompt exists
• Fixed Device cannot upgrade to OEM version firmware
• Fixed Device web page displays abnormally on Programmable Keys page and GDS opendoor page
• Fixed Automatic redial will not be triggered after call timeout
• Fixed Device LCD does not display SRTP icon in talking when establishing call with non-registered account line key
• Fixed Offhook auto dial does not work after the phone is locked
• Fixed time display on the LCD screen is wrong
• Fixed The time zone on LCD is not consistent with the time zone on the web UI
• Fixed PC port VLAN default values of LCD and Web are inconsistent
• Fixed Keypad lock fails after automatic redial is enabled
• Fixed Device can still dial the non-emergency number after the keypad is locked
• Fixed Device crashes after emptying emergency numbers
• Fixed Device does not prompt authentication for CFGmac.xml that sets authentication
• Fixed The keypad lock timer does refresh after the keypad lock timer is set to 0
• Fixed the Internet Protocol on LCD and Web are inconsistent
• Fixed Off-hook auto dial feature works abnormally
• Fixed Device freezes after uploading CA certificate of about 3M
• Fixed Nonce Count is wrong for Re-Register if there is 401 challenge for Invite
• Fixed Bypass Dial plan configuration causes the failure on network Conference and voicemail calling
• Fixed Device does not display the correct Issued By and Expiration from the loaded CA certificate
• Fixed Device crashes after doing search in the directory LDAP
• Fixed Device taking more than 1 second to respond the second INVITE
• Fixed Device displays only the first contact's number when one contact includes multiple numbers
• Fixed Device shows incorrect display under local call Log
• Fixed GXP1630 account 3 does not have option "Callee ID display"
• Fixed Calls don't disconnect when using TLS
• Fixed Device scrolling feature on dialed calls logs not behaving correctly
• Fixed 802.1X port based authentication not working
• Fixed Phonebook screen displayed extra Sort softkey after moving cursor to LDAP Directory
• Fixed Device sends anonymous identity for Radius authentication
• Fixed Delay on switch during Device booting
• Fixed Device does not send Custom Certificate for TLS Authentication
• Fixed Device Jitter Buffer Type does not work correctly
• Fixed Device freezes randomly
• Fixed Device does not display lock sign when the call is established using both SRTP and TLS

NEW FEATURES OVERVIEW
This section lists major new features and describes how to use it from the user’s point of view.
MORE SPECIAL CHARACTERS VIA KEYPAD

• Functionality

Now users are able to input pound and plus sign through keypad in SIP account.

SCROLL LONG NAMES

• Functionality

Now long names will scroll though LCD to display all the contents instead of cutting off in the middle of the screen.

GNU GPL LICENSE

• Functionality

Now GNU GPL License information will be displayed on device’s Web UI—Status—GNU GPL License, and LCD menu.

PROVISION VPK/MPK

• Web Configuration

This option can be found under device web UI → Settings → Programmable Keys → Programmable Keys—Mode.

uyên

• Functionality

Now users are able to use VPK/MPK to trigger provision process.

BROADSOFT DIRECTORIES

• Functionality

Now the users can enter login credentials through LCD for Broadsoft Directories.
SIP USER ID FIELD IMPROVEMENT

- **Functionality**

Now users are able to register an account with a SIP user ID that carries "@". For example, you can put "111@test.com" in your SIP User ID field. When registering, the phone will register your account as "111@test.com" instead of 111.

Note: The server domain will not be included in the SIP from header.

SCREEN CAPTURE ACTION URI

- **Functionality**

Now users can use URI to capture DUT screen. Format is as below:

(http://[IP_Address]/cgi-bin/api-capture_screen?passcode=[Passcode])

ADD AUTH HEADER ON INITIAL REGISTER

- **Web Configuration**

This option can be found under device web UI -> Account -> Account # -> SIP Settings -> Basic Settings.

- **Functionality**

When this option is enabled, Device will send Authentication Header on the first Register instead of second Register.

- **Pvalue**

<table>
<thead>
<tr>
<th>Pvalue</th>
<th>Description</th>
<th>Value Range</th>
<th>Default</th>
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</thead>
<tbody>
<tr>
<td>P2359</td>
<td>&lt;New Pvalue&gt; Add Auth Header On Initial REGISTER</td>
<td>Value = 0; No Value = 1; Yes</td>
<td>No</td>
</tr>
</tbody>
</table>

HOST NAME(OPTION 12)

- **Web Configuration**

This option can be found under device web UI -> Network -> Basic Settings.

- **Functionality**

Now users can add host name for their DHCP INFORM using Option 12.
- **Pvalue**

<table>
<thead>
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<th>Value Range</th>
<th>Default</th>
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<tbody>
<tr>
<td>P146</td>
<td>&lt;New Pvalue&gt; Host Name</td>
<td>String</td>
<td>N/A</td>
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FIRMWARE VERSION 1.0.4.140

PRODUCT NAME
GXP1610, GXP1615, GXP1620, GXP1625, GXP1628 and GXP1630

DATE
07/19/2019

FIRMWARE FILE INFORMATION

- Firmware file name: gxp1600fw.bin
  MD5: 299f43fc085fe6732ec4dea894153818

ENHANCEMENT

- Added Support for Emergency call under functional lock [FUNCTIONAL LOCK]
- Added Support for Ukrainian language and input
- Added Support for callee ID display [CALLEE ID DISPLAY]
- Added ability to create Custom Call Screen Softkey Layout [CALLEE ID DISPLAY]
- Added ability to customize the feature code [CUSTOM CALL FEATURE CODE]
- Added option to specify which certificate to trust [LOAD CA CERTIFICATES]
- Added show both call session timer and hold duration timer on LCD during call hold [SHOW ON HOLD DURATION]
- Added more custom ringtone choices [INCREASE CUSTOM RINGTONE TO 10 RINGTONES]
- Added option to disable # key from acting as redial key [ENABLE/DISABLE USING # FOR REDIAL]
- Added support for variable $active_url in Action URL [SUPPORT $active_url IN ACTION URL]
- Added upload button to upload SIP TLS Private Key [UPLOAD FUNCTION FOR SIP TLS CERTIFICATE/PRIVATE KEY]
- Added option to hide reject softkey on ringing state [ENABLE/DISABLE REJECT SOFTKEY]
- Added option "Disable Busy Tone on Remote Disconnect" [ENABLE/DISABLE BUSY TONE ON REMOTE DISCONNECT]
- Added option to specify DNS SRV query frequency [SUPPORT DNS CACHE REFRESH TIME]
- Added ability to open door in idle stage for GDS integration [SUPPORT OPEN GDS DOOR IN IDLE STAGE]
- Added ability to use special characters on 802.1X MD5 password under device web UI → Network → Advanced Settings
- Added option to choose CID display behavior (static or dynamic) [CALLER ID DISPLAY MODE]
- Added option to enable/disable Diversion info display [ENABLE/DISABLE DIVERSION INFORMATION DISPLAY]
• Changed default value for “Enable CDP” to be “Enabled” instead of “Disabled” under web UI → Network → Advanced Settings
• Added options to disable hook switch [DISABLE HOOK SWITCH]

BUG FIX
• Fixed Wrong caller issue when AoC is used
• Fixed Cyrillic names does not scroll on call screen
• Fixed Device doesn’t show the display name on the missed call log if the phone is busy when the call is received
• Fixed Custom alert info for auto answer does not work
• Fixed Device cannot establish SRTP call when SDP include media for both RTP and SRTP
• Fixed Multicast Paging channel issue
• Fixed Caller ID name not scrolling in LCD when incoming call in other language except English
• Fixed "$active_user" variable not working for SIP Registration/Un-registration
• Fixed Phone does not perform IP call when configuring a VPK with speed dial
• Fixed Phone crashes when modifying the Anonymous Call Rejection setting
• Fixed Azores time zone does not include DST definition
• Fixed Phonebook from 1.0.4.67 can't upload to 1.0.4.128
• Fixed Phone replies with 200 OK "sendrecv" upon receiving INVITE "sendonly" for SIP/TLS or SIP/UDP with Broadsoft
• Fixed Phone doesn't trigger the provisioning when the DNS resolution is made with the secondary DNS server
• Fixed Security related issues
• Fixed Character "+" is not supported for phonebook contacts added manually in the Web GUI
• Fixed Eventlist BLF not working properly
• Fixed Device does not send PRACK in Allow header when send '180 Ringing'
• Hide System Softkey on Idle Page still show Redial
• Fixed Call established without audio randomly
• Fixed When both lines have line mode set to LINE, the same account shows duplicate for Line 1 and Line 2 (for GXP1610/1615 which only supports only 1 SIP account)
• Fixed Phone can't change the ringtone from the keypad

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

FUNCTIONAL LOCK
• Web Configuration

This option can be found under device web UI → Maintenance → Security Settings → Security.
• **Functionality**

Now users can set Keypad Functional Lock to their phone instead of completely locking the phone. To enable keypad lock, simply set the option "Enable STAR Key Keypad Locking" to “Yes” and long press STAR Key("*" ) to enable keypad lock. Using Keypad Lock Type to change from full lock to Functional. Functional lock will disable all the function keys on the keypad except for emergency calls. This ensure users can still call out on emergency.

To set emergency number, input number separated by ",” in the emergency number field (i.e. 911, 110, 112). If the user forgets to lock their phone and do not want others to use their phone, they can set the “Keypad Lock Timer” for the phone. Once the timer is up, the phone will automatically get locked.

• **New Pvalue**

<table>
<thead>
<tr>
<th>Pvalue</th>
<th>Description</th>
<th>Value Range</th>
<th>Default</th>
</tr>
</thead>
<tbody>
<tr>
<td>P1382</td>
<td>Enable STAR Key Keypad Locking</td>
<td>Value=0; No Value=1; Yes</td>
<td>No</td>
</tr>
<tr>
<td>P8383</td>
<td>Keypad Lock Type</td>
<td>Value=0; Full Value=1; Functional</td>
<td>Full</td>
</tr>
<tr>
<td>P1383</td>
<td>Password to Lock/Unlock</td>
<td>String</td>
<td>N/A</td>
</tr>
<tr>
<td>P8385</td>
<td>Emergency Numbers</td>
<td>String</td>
<td>N/A</td>
</tr>
<tr>
<td>P8384</td>
<td>Keypad Lock Timer</td>
<td>0-3600</td>
<td>0</td>
</tr>
</tbody>
</table>

**CALLEE ID DISPLAY**

• **Web Configuration**

This option can be found under device web UI → Accounts → Account # → SIP Settings → Basic Settings → Callee ID Display.
• **Functionality**

Now device supports Callee ID display. When the option is set to "Auto", the phone will update the callee ID in the order of P-Asserted Identity Header, Remote-Party-ID Header and To Header in the 180 Ringing. When set to "Disabled", callee ID will be displayed as "Unavailable". When set to "To Header", caller ID will not be updated and displayed as To Header.

• **Pvalue**

<table>
<thead>
<tr>
<th>Pvalue</th>
<th>Description</th>
<th>Value Range</th>
<th>Default</th>
</tr>
</thead>
<tbody>
<tr>
<td>P26025</td>
<td>Callee ID Display</td>
<td>Value=0; Auto Value=1; Disable Value=2; To Header</td>
<td>Auto</td>
</tr>
</tbody>
</table>

**CUSTOM CALL FEATURE CODE**

• **Web Configuration**

This option can be found under device web UI → Accounts → Account x → Feature Codes.
• **Functionality**

Users can customize the feature codes for DND and call forwarding rules. When local feature is disabled, the custom feature code will take effect for DND and call forwarding rules as configured.

• **Pvalue**

<table>
<thead>
<tr>
<th>Pvalue</th>
<th>Description</th>
<th>Value Range</th>
<th>Default</th>
</tr>
</thead>
<tbody>
<tr>
<td>P26013</td>
<td>&lt;New Pvalue&gt; Account 1: Call Forward Unconditionally (All) - On</td>
<td>String</td>
<td></td>
</tr>
<tr>
<td>P26014</td>
<td>&lt;New Pvalue&gt; Account 1: Call Forward Unconditionally (All) - Off</td>
<td>String</td>
<td></td>
</tr>
<tr>
<td>P26009</td>
<td>&lt;New Pvalue&gt; Account 1: Call Forward Busy - On</td>
<td>String</td>
<td></td>
</tr>
<tr>
<td>P26010</td>
<td>&lt;New Pvalue&gt; Account 1: Call Forward Busy - Off</td>
<td>String</td>
<td></td>
</tr>
<tr>
<td>P26011</td>
<td>&lt;New Pvalue&gt; Account 1: Call Forward Delayed (No Answer) - On</td>
<td>String</td>
<td></td>
</tr>
<tr>
<td>P26012</td>
<td>&lt;New Pvalue&gt; Account 1: Call Forward Delayed (No Answer) - Off</td>
<td>String</td>
<td></td>
</tr>
<tr>
<td>P2444</td>
<td>&lt;New Pvalue&gt; Account 2: Do Not Disturb (DND) - On</td>
<td>String</td>
<td></td>
</tr>
<tr>
<td>P2445</td>
<td>&lt;New Pvalue&gt; Account 2: Do Not Disturb (DND) - Off</td>
<td>String</td>
<td></td>
</tr>
<tr>
<td>P26113</td>
<td>&lt;New Pvalue&gt; Account 2: Call Forward Unconditionally (All) - On</td>
<td>String</td>
<td></td>
</tr>
<tr>
<td>P26114</td>
<td>&lt;New Pvalue&gt; Account 2: Call Forward Unconditionally (All) - Off</td>
<td>String</td>
<td></td>
</tr>
<tr>
<td>P26109</td>
<td>&lt;New Pvalue&gt; Account 2: Call Forward Busy - On</td>
<td>String</td>
<td></td>
</tr>
<tr>
<td>P26110</td>
<td>&lt;New Pvalue&gt; Account 2: Call Forward Busy - Off</td>
<td>String</td>
<td></td>
</tr>
</tbody>
</table>
LOAD CA CERTIFICATES

- Web Configuration

A new option "Load CA Certificate" can be found under device web UI → Maintenance → Security Settings → Trusted CA Certificates.
• **Functionality**

Users can specify which CA certificate the phone needs to trust.

- **Default Certificates**: The phone will verify the server certificate based on the built-in trusted certificates list.
- **Custom Certificates**: The phone will verify the server certificate based on the custom trusted certificates list.
- **All Certificates**: The phone will verify the server certificate based on the trusted certificates list including build-in and custom trusted certificates.

• **Pvalue**

<table>
<thead>
<tr>
<th>Pvalue</th>
<th>Description</th>
<th>Value Range</th>
<th>Default</th>
</tr>
</thead>
<tbody>
<tr>
<td>P8502</td>
<td>&lt;New Pvalue&gt; Load CA Certificates</td>
<td>Value = 0; Default Certificates</td>
<td>2</td>
</tr>
<tr>
<td></td>
<td></td>
<td>Value = 1; Custom Certificates</td>
<td></td>
</tr>
<tr>
<td></td>
<td></td>
<td>Value = 2; All Certificates</td>
<td></td>
</tr>
</tbody>
</table>

**SHOW ON HOLD DURATION**

• **Web Configuration**

A new option “Show On Hold Duration” can be found under device web UI ➔ Settings ➔ Call Features.
• Functionality

When “Show On Hold Duration” is enabled, the Hold duration timer will be displayed on the LCD if the call is put on hold.

• Pvalue

<table>
<thead>
<tr>
<th>Pvalue</th>
<th>Description</th>
<th>Value Range</th>
<th>Default</th>
</tr>
</thead>
<tbody>
<tr>
<td>P8503</td>
<td>&lt;New Pvalue&gt; Show On Hold Duration</td>
<td>Value = 0; No</td>
<td>0</td>
</tr>
<tr>
<td></td>
<td></td>
<td>Value = 1; Yes</td>
<td></td>
</tr>
</tbody>
</table>

INCREASE CUSTOM RINGTONE TO 10 RINGTONES

• Web Configuration

The related options can be found under device web UI → Settings → Preferences → Ringtone, and also in Account Ringtone settings under web UI → Accounts → Account x → Call Settings.
- “Default Ringtone” now supports up to 10 custom ringtones to be selected.
- “Total Number of Custom Ringtone Update” allows user to configure up to 10 custom ringtones.

• Functionality

The phone now can download up to 10 different custom ringtones.
ENABLE/DISABLE USING # FOR REDIAL

- **Web Configuration**

A new option “Use Pound (#) For Redial” can be found under device web UI ➔ Settings ➔ Call Features.

![Use Pound (#) For Redial](image)

Also, the previous option “Use # as Dial key” under web UI ➔ Accounts ➔ Account x ➔ Call Settings is now renamed to “Key as Send” with option “Disabled” and “Pound (#)”.

![Key as Send](image)

- **Functionality**

When “Use Pound (#) For Redial” is set “No”, the pound key # will not act as redial when it’s pressed in idle screen. Also, if “Key As Send” is set to “Pound (#)”, when entering a number first and then pressing pound (#) key, the phone will dial out immediately.

- **Pvalue**

<table>
<thead>
<tr>
<th>Pvalue</th>
<th>Description</th>
<th>Value Range:</th>
<th>Default</th>
</tr>
</thead>
<tbody>
<tr>
<td>P8504</td>
<td>&lt;New Pvalue&gt; Use Pound (#) For Redial</td>
<td>Value = 0; No Value = 1; Yes</td>
<td>1</td>
</tr>
<tr>
<td>P72</td>
<td>Account 1: Key as Send</td>
<td>Value = 0; Disabled Value = 1; Pound (#)</td>
<td>1</td>
</tr>
<tr>
<td>P492</td>
<td>Account 2: Key as Send</td>
<td>Value = 0; Disabled Value = 1; Pound (#)</td>
<td>1</td>
</tr>
<tr>
<td>P592</td>
<td>Account 3: Key as Send</td>
<td>Value = 0; Disabled Value = 1; Pound (#)</td>
<td>1</td>
</tr>
</tbody>
</table>

**SUPPORT $active_url IN ACTION URL**

Under device web UI ➔ Maintenance ➔ Action URL, if an action field is configured with URL including...
active_uri=$active_uri, the output URI will contain the actual SIP URI.

Assuming the device has the SIP account configured with the following:
- Account Name: Test
- SIP Server: sip.grandstream.com
- SIP User ID: 1002

<table>
<thead>
<tr>
<th>WebUI Configuration: Maintenance → Action URL page</th>
<th>Action</th>
<th>Output URL</th>
</tr>
</thead>
</table>
| Incoming Call: http://192.168.5.224:8080/in=$active_uri | Make incoming call to the device | "GET /in=1002@sip.grandstream.com HTTP/1.1"
| Outgoing Call: http://192.168.5.224:8080/out=$active_uri | Make outgoing call from the device | "GET /out=1002@sip.grandstream.com HTTP/1.1"
| Unregistered: http://192.168.5.224:8080/unreg=$active_uri | Unregister the SIP account on the device | "GET /unreg=1002@sip.grandstream.com HTTP/1.1"
| Registered: http://192.168.5.224:8080/reg=$active_uri | Register SIP account on the device | "GET /reg=1002@sip.grandstream.com HTTP/1.1"

**UPLOAD FUNCTION FOR SIP TLS CERTIFICATE/PRIVATE KEY**

Under device web UI → Maintenance → Security Settings → Security, users can now use the “Upload” and “Delete” buttons to manage SIP TLS Certificate and SIP TLS Private Key.
ENABLE/DISABLE REJECT SOFTKEY

- **Web Configuration**

A new option “Hide/Disable REJECT Key” can be found under device web UI → Settings → Call Features.

![Hide/Disable REJECT Key](image)

- **Functionality**

When “Hide/Disable REJECT Key” is set to “Yes”, the device will not display REJECT softkey upon incoming call.

- **Pvalue**

<table>
<thead>
<tr>
<th>Pvalue</th>
<th>Description</th>
<th>Value Range</th>
<th>Default</th>
</tr>
</thead>
<tbody>
<tr>
<td>P8511</td>
<td>&lt;New Pvalue&gt;</td>
<td>Value = 0; No&lt;br&gt;Value = 1; Yes</td>
<td>0</td>
</tr>
<tr>
<td></td>
<td>Hide/Disable REJECT Key</td>
<td></td>
<td></td>
</tr>
</tbody>
</table>

ENABLE/DISABLE BUSY TONE ON REMOTE DISCONNECT

- **Web Configuration**

A new option “Disable Busy Tone on Remote Disconnect” can be found under device web UI → Settings → Call Features.

![Disable Busy Tone on Remote Disconnect](image)

- **Functionality**

When “Disable Busy Tone on Remote Disconnect” is set to “Yes”, the device will not hear the busy tone after remote party hangs up the call.
SUPPORT DNS CACHE REFRESH TIME

**Web Configuration**

A new option “DNS Cache Refresh Time” can be found under device web UI → Settings → General Settings.

![DNS Cache Refresh Time](image)

**Functionality**

The phone will send out DNS queries based on the configured DNS Cache Refresh Time after responding to refresh IP addresses for SIP.

**Pvalue**

<table>
<thead>
<tr>
<th>Pvalue</th>
<th>Description</th>
<th>Value Range</th>
<th>Default</th>
</tr>
</thead>
<tbody>
<tr>
<td>P4208</td>
<td>&lt;New Pvalue&gt; DNS Cache Refresh Time</td>
<td>5 – 1440, in minutes</td>
<td>30</td>
</tr>
</tbody>
</table>

SUPPORT OPEN GDS DOOR IN IDLE STAGE

**Web Configuration**

Assuming the device already has GDS open door setting configured under device web UI → Settings → External Service, navigate to device web UI → Settings → Programmable Keys, add an MPK or Line key, choose “GDS OpenDoor” for key “Mode” and enter the GDS SIP extension for the MPK/Line key value.
Note:
The GDS device must be properly configured as well such as enabling “Remote PIN To Open Door” under GDS web UI->Door System Settings->Basic Settings.

- **Functionality**

When the device is in idle stage, pressing the configured MPK or Line key will open the door via GDS.

**CALLER ID DISPLAY MODE**

- **Web Configuration**

A new option “Caller ID Display” can be found under device web UI-> Settings -> Preferences -> LCD Display.

- **Functionality**

Users can configure “Caller ID Display Mode” to control the caller ID display preference:
  - Dynamic: If the caller ID is long, it will keep scrolling until it’s displayed completed.
  - Static: The caller ID will not scroll.

- **Pvalue**

<table>
<thead>
<tr>
<th>Pvalue</th>
<th>Description</th>
<th>Value Range</th>
<th>Default</th>
</tr>
</thead>
<tbody>
<tr>
<td>P8473</td>
<td>Caller ID Display Mode</td>
<td>Value = 0; Dynamic Value = 1; Static</td>
<td>0</td>
</tr>
</tbody>
</table>

**ENABLE/DISABLE DIVERSION INFORMATION DISPLAY**

- **Web Configuration**
A new option “Enable Diversion Information Display” can be found under device web UI → Settings → Call Features.

- **Functionality**

When “Enable Diversion Information Display” is set to “Yes”, Diversion info will be displayed if the device receives INVITE with Diversion header.

Example:
For each SIP account that has “Remote Info Display” setting under web UI → Accounts → Account x → Call Settings, it will follow below behaviors when “Enable Diversion Information Display” is enabled and when “Remote Info Display” is set to “Name and Number”:
- If A calls B and B has the call forwarded to the GXP16xx, the GXP16xx will display “A_name A_number (Diverted From B_number)” on LCD. If the GXP16xx’s local phonebook has B in it, B_name will be displayed instead of B_number in the “Diverted From” part.
- If A calls B, B answers and then B transfers to the GXP16xx, the GXP16xx will display “A_name A_number (Transferred From B_number)” on LCD. If the GXP16xx’s local phonebook has B in it, B_name will be displayed instead of B_number in the “Transferred From” part.

- **Pvalue**

<table>
<thead>
<tr>
<th>Pvalue</th>
<th>Description</th>
<th>Value Range</th>
<th>Default</th>
</tr>
</thead>
<tbody>
<tr>
<td>P22237</td>
<td>Enable Diversion Information Display</td>
<td>Value = 0; No Value = 1; Yes</td>
<td>1</td>
</tr>
</tbody>
</table>

**DISABLE HOOK SWITCH**

- **Web Configuration**

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

- **Functionality**
Now the users can choose the behavior for hook switch for the phone’s handset.

- If set to “No”, the phone will answer the call when the handset is off-hook. This is the default behavior.
- If set to “Yes”, the hook switch will not work to trigger on hook/off hook at all. The handset audio channel will not work during phone call either.
- If set to “For Answering Call”, the phone’s hook switch will not work to answer incoming call. Users will need press softkey or speaker/headset/line hard key to answer the call. The hook switch will still work to switch audio channel to handset. For example, if the incoming call is answered by pressing ANSWER softkey, users can lift handset (off-hook) to switch audio channel to handset.

**Pvalue**

<table>
<thead>
<tr>
<th>Pvalue</th>
<th>Description</th>
<th>Value Range</th>
<th>Default</th>
</tr>
</thead>
<tbody>
<tr>
<td>P6763</td>
<td>&lt;New Pvalue&gt; Disable Hook Switch</td>
<td>Value = 0; No Value = 1; Yes Value = 2; For Answering call</td>
<td>Value = 0; No</td>
</tr>
</tbody>
</table>
FIRMWARE VERSION 1.0.4.132

PRODUCT NAME
GXP1610, GXP1615, GXP1620, GXP1625, GXP1628 and GXP1630

DATE
10/18/2018

FIRMWARE FILE INFORMATION
- Firmware file name: gxp1600fw.bin
- MD5: 87e6a4abb8d1166106f7176fe2d96dc3

ENHANCEMENT
- Added Support for the string "tftp://" inside DHCP Option 66 [TFTP:// SUPPORT]
- Increased phonebook limit to 1000
- Updated German Web Language
- Updated Portuguese Web Language
- Updated Portuguese LCD Language
- Updated French Web Language
- Updated French LCD Language
- Updated Spanish Web Language
- Updated Spanish LCD Language

BUG FIX
- Fixed Timezone values not correct for some locations
- Fixed SSH Public Key is invalid
- Fixed Bypass dial plan configuration causes the failure on network Conference and voicemail calling
- Fixed Device crashes randomly and generates core dump
- Fixed Device crashed after setting account active to yes
- Fixed Access to a root shell (CVE-2018-17565)
- Fixed Remote manipulation of some nvram parameters without authorization (CVE-2018-17564)
- Fixed Remote dump of all the nvram parameters without authorization (CVE-2018-17563)

NEW FEATURES OVERVIEW
This section lists major new features and describes how to use it from the user’s point of view.
TFTP:// SUPPORT

• Functionality

Previously, tftp protocol will be forced to add into the values you input. For example, if you have “192.168.1.1” in the field, the phone will send tftp request to “tftp://192.168.1.1”. However, if you have “tftp://192.168.1.1” in option 66, phone will send “tftp://tftp://192.168.1.1”. Now with the new enhancement, when setting up DHCP option 66, users will not need to care about the protocol. You can input either “192.168.1.1” or “tftp://192.168.1.1”. Both will work as expected.
FIRMWARE VERSION 1.0.4.128

PRODUCT NAME
GXP1610, GXP1615, GXP1620, GXP1625, GXP1628 and GXP1630

DATE
08/03/2018

FIRMWARE FILE INFORMATION
• Firmware file name: gxp1600fw.bin

MD5: 2acb6b264795f134f6968d01a6c83c1f

ENHANCEMENT
• Added Support for SNMPv3
• Added FTP(FTPS) support for provisioning and firmware upgrade [FTP (FTPS) SUPPORT]
• Improved delay on switch when device is booting
• Added Support to configure device with custom certificate signed by custom CA certificate [CUSTOM CERTIFICATE]
• Added Support for local pcap capture [LOCAL PCAP CAPTURE]
• Added Support for Group Listening softkey [GROUP LISTENING]
• Added Support for more Action URL notification [SUPPORT MORE ACTION URL NOTIFICATION]
• Added Support for XML Application [XML APPLICATIONS]
• Added Support for RTCP-XR
• Added Broadsoft call park notification feature
• Added Codec Negotiation Priority feature [CODEC NEGOTIATION PRIORITY]
• Added Support for "tftp://" in DHCP Option 66
• Added Support for more characters in password input from LCD [MORE CHARACTERS FOR PASSWORD INPUT FROM LCD]
• Added Support to customize idle time to logout the web access [WEB SESSION TIMEOUT]
• Added Support to customize number of failed attempt to access web GUI [WEB ACCESS ATTEMPT LIMIT]
• Added Support for BLF call pick-up to go through dial plan [BYPASS DIAL PLAN]
• Added Support of on hook dialing with Live DialPad [LIVE DIALPAD]
• Added Support for CID and Image from SIP NOTIFY [CID AND IMAGE FROM SIP NOTIFY]
• Added Support to randomize TR069 Inform Interval [RANDOMIZED TR069 STARTUP]
• Added Support to configure off-hook auto dial delay [OFFHOOK AUTO DIAL DELAY]
• Added Support for auto answer delay [AUTO ANSWER DELAY]
• Added More fields for Distinctive Ring Tone (Matching Rule) [MORE MATCHING RULES SUPPORTED]
• Add Support to enable or disable Acoustic Echo Cancellation [ENABLE ENHANCED ACOUSTIC ECHO CANCELLER]
• Add Support to download certificate files during provisioning [DOWNLOAD CERTIFICATE FILES DURING PROVISION]
• Add Support of on hold reminder tone [ON HOLD REMINDER TONE]

**BUG FIX**

• Fixed Certain path syntax can cause device to send config request multiple times and break firmware provision
• Fixed Device doesn't require the old GUI Admin password to define a new password from the LCD
• Fixed Option 66 provisioning not sending authentication credentials
• Fixed HTTP provision will return 505 error if config path contains spaces
• Disabled 802.1Q tag when VLAN ID is 0 and priority value is different to 0
• Fixed Device does not re-establish PPPoE connection after terminating request from server
• Fixed Upgrade web UI page spelling mistakes
• Fixed Device only encrypt signal but not RTP under the CBM model
• Fixed Device LCD updates network status slowly after factory reset
• Fixed Device cannot upgrade firmware successfully with dummy file break provision process with Broadsoft
• Fixed Device Eventlist BLF not updating keys correctly with Broadsoft
• Fixed Device priority alert silent alerting not working with Broadsoft
• Fixed Some Pvalues get abnormal or wrong after factory reset
• Fixed Device gets freezing after downloading an empty phonebook
• Fixed Device keeps rebooting all the time
• Fixed Device doesn't display the correct CID in call history after *code call pick-up
• Fixed Daylight-saving time does not work correctly for some time zones
• Fixed Device crashes after receiving instant message
• Fixed Device crashes after enabling account 2 and setting MPK as BLF or presence watcher
• Fixed Device does not send SIP SUBSCRIBE messages when switch between OBPs
• Fixed Device hangs up the call after sending call park feature code
• Fixed INVITE cannot be print out in syslog
• Fixed The tooltip of "Proxy server Switch Delay" display error
• Updated Korean Translations
• Fixed Device still sends TCP Keep-Alive with all Keep-Alive settings off
• Fixed Device only CONTACT header is using SIPS when setting "SIP URI Scheme when using TLS" to SIPS
• Fixed Device still sends DEBUG, WARNING and INFO when set syslog Level to "ERROR"
• Fixed Device "Download and Process ALL Available Config Files" works abnormally
• Fixed Daylight Saving Time displayed incorrectly on LCD
• Removed A reboot request when a timezone p-value is updated
• Updated BLF State to follow pval value
• Increased Field to display incoming caller ID
• Changed Backup proxy SIP REGISTRATION behavior for iPageOn
• Updated Call hold is not allowed if receiving a 488 for iPageOn
• Fixed Device one-way audio issue after semi-attended transfer when using SRTP for iPageOn
• Fixed BLF stays red issue when monitored extension receives more than one call for PortaOne
• Fixed Configuration backup should not include CA/Custom certificate Pvalue
• Fixed Echo issue on speaker
• Fixed Multicast Paging Priority disable option does not working
• Fixed Device intermittently not allowing terminal to be answered
• Fixed Device IP is shown as 0.0.0.0
• Fixed Click-To-Dial feature does not bypass the dial plan rules
• Fixed Customer certificate provision does not work with certain Pvalue
• Fixed ATE-IP call test failed
• Changed SSRC/SEQ number in outgoing RTP packets between early media and main media
• Updated phonebook string in LCD and Web menu

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

FTP (FTPS) SUPPORT
• Web Configuration

This option can be found under device web UI ➔ Maintenance ➔ Upgrade and Provisioning ➔ Config Upgrade via.

<table>
<thead>
<tr>
<th>Config Upgrade via</th>
<th>TFTP</th>
<th>HTTP</th>
<th>HTTPS</th>
<th>FTP</th>
<th>FTPS</th>
</tr>
</thead>
<tbody>
<tr>
<td>Firmware Upgrade via</td>
<td>TFTP</td>
<td>HTTP</td>
<td>HTTPS</td>
<td>FTP</td>
<td>FTPS</td>
</tr>
</tbody>
</table>

• Functionality

Now other than TFTP/HTTP/HTTPS, we also support FTP and FTPS for provisioning and firmware upgrade.

• Pvalue

<table>
<thead>
<tr>
<th>Pvalue</th>
<th>Description</th>
<th>Value Range</th>
<th>Default</th>
</tr>
</thead>
<tbody>
<tr>
<td>P212</td>
<td>Config Upgrade via</td>
<td>Value = 0; TFTP</td>
<td>2</td>
</tr>
<tr>
<td></td>
<td></td>
<td>Value = 1; HTTP</td>
<td></td>
</tr>
<tr>
<td>Value</td>
<td>Description</td>
<td></td>
<td></td>
</tr>
<tr>
<td>-------</td>
<td>-------------</td>
<td></td>
<td></td>
</tr>
<tr>
<td>2</td>
<td>HTTPS</td>
<td></td>
<td></td>
</tr>
<tr>
<td>3</td>
<td>FTP</td>
<td></td>
<td></td>
</tr>
<tr>
<td>4</td>
<td>FTPS</td>
<td></td>
<td></td>
</tr>
</tbody>
</table>

### CUSTOM CERTIFICATE

- **Web Configuration**

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

![Custom Certificate](image)

- **Functionality**

Now if the users want to upload their own certificate, they can upload from the custom certificate that allows the server to verify with the phone.

### LOCAL PCAP CAPTURE

- **Web Configuration**

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

![Packet Capture](image)

- **Functionality**

Now if users want to capture their phone’s traffic, they can use the packet capture function on the phone. After pressing start, you’ll see:

![Packet Capture](image)

This means the capture is running. If you want to stop capturing, just press Stop. Then you’ll be able to download the pcap file from the phone.

### GROUP LISTENING

- **Web Configuration**
This option can be found under device web UI → Settings → Preferences → Audio Control.

Enable Group Listening

Group Listen with Speaker

- **Functionality**

Enabling Group Listening will enable speaker sound while the handset/headset still works as they should.

- **Pvalue**

<table>
<thead>
<tr>
<th>Pvalue</th>
<th>Description</th>
<th>Value Range</th>
<th>Default</th>
</tr>
</thead>
<tbody>
<tr>
<td>P22100</td>
<td>&lt;New Pvalue&gt; Enable Group Listening</td>
<td>Value = 0; No; Value = 1; Yes</td>
<td>No</td>
</tr>
<tr>
<td>P22163</td>
<td>&lt;New Pvalue&gt; Group Listen with Speaker</td>
<td>Value = 0; No; Value = 1; Yes</td>
<td>No</td>
</tr>
</tbody>
</table>

**SUPPORT MORE ACTION URL NOTIFICATION**

- **Web Configuration**

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

- **Functionality**

Now device supports more action URL, including Unconditional Forward, No Answer Forward, Transfer, Mute, etc.

**XML APPLICATIONS**

- **Web Configuration**

This option can be found under device web UI → Settings → XML Application.

<table>
<thead>
<tr>
<th>XML Application</th>
</tr>
</thead>
<tbody>
<tr>
<td>Server Path</td>
</tr>
<tr>
<td>Softkey Label</td>
</tr>
<tr>
<td>XML Service</td>
</tr>
</tbody>
</table>

- **Functionality**
This feature allows users to access XML service information in web server, and to present the information in LCD, also to customize screen display.

- **Pvalue**

<table>
<thead>
<tr>
<th>Pvalue</th>
<th>Description</th>
<th>Value Range</th>
<th>Default</th>
</tr>
</thead>
<tbody>
<tr>
<td>P340</td>
<td>Idle Screen XML Download</td>
<td>Value = 0; Disabled</td>
<td>Disabled</td>
</tr>
<tr>
<td></td>
<td></td>
<td>Value = 1; Enabled, Use TFTP</td>
<td></td>
</tr>
<tr>
<td></td>
<td></td>
<td>Value = 2; Enabled, HTTP</td>
<td></td>
</tr>
<tr>
<td></td>
<td></td>
<td>Value = 3; Enabled, HTTPS</td>
<td></td>
</tr>
<tr>
<td>P1349</td>
<td>Download Screen XML at Boot-up</td>
<td>Value = 0; No</td>
<td>No</td>
</tr>
<tr>
<td></td>
<td></td>
<td>Value = 1; Yes</td>
<td></td>
</tr>
<tr>
<td>P1343</td>
<td>Use Custom Filename</td>
<td>Value = 0; No</td>
<td>No</td>
</tr>
<tr>
<td></td>
<td></td>
<td>Value = 1; Yes</td>
<td></td>
</tr>
<tr>
<td>P341</td>
<td>Idle Screen XML Server Path</td>
<td>String</td>
<td>N/A</td>
</tr>
<tr>
<td>P337</td>
<td>Server Path</td>
<td>String</td>
<td>N/A</td>
</tr>
<tr>
<td>P352</td>
<td>Softkey Label</td>
<td>String</td>
<td>XML Service</td>
</tr>
</tbody>
</table>

### CODEC NEGOTIATION PRIORITY

- **Web Configuration**

  This option can be found under device web UI → Account → Account # → Audio Settings → Codec Negotiation Priority.

- **Functionality**

  Now the users can decide which codec phone will use as priority.

- **Pvalue**

<table>
<thead>
<tr>
<th>Pvalue</th>
<th>Description</th>
<th>Value Range</th>
<th>Default</th>
</tr>
</thead>
<tbody>
<tr>
<td>P29061</td>
<td>&lt;New Pvalue&gt;</td>
<td>Value = 0; Caller</td>
<td>Callee</td>
</tr>
<tr>
<td>P29161</td>
<td>Codec Negotiation Priority</td>
<td>Value = 1; Callee</td>
<td></td>
</tr>
</tbody>
</table>
MORE CHARACTERS FOR PASSWORD INPUT FROM LCD

- **Functionality**

  Now users can enter more characters from the LCD screen including "<, >, ?, {, }, [], ^" these characters will be allow to input by using number key 1.

WEB SESSION TIMEOUT

- **Web Configuration**

  This option can be found under device web UI → Maintenance → Security Settings → Security → Web Session Timeout.

  ![Web Session Timeout](image)

  - **Functionality**

    Now users can set how long they want to keep the web UI to stay idle before the web log them out.

  - **Pvalue**

    | Pvalue   | Description             | Value Range | Default |
    |----------|-------------------------|-------------|---------|
    | P28116   | <New Pvalue> Web Session Timeout | 2-60        | 10      |

WEB ACCESS ATTEMPT LIMIT

- **Web Configuration**

  This option can be found under device web UI → Maintenance → Security Settings → Security → Web Access Attempt Limit.

  ![Web Access Attempt Limit](image)

  - **Functionality**

    Now users can set how many times they want their password to be tested before they lock the IP out for some time.

  - **Pvalue**

    | Pvalue | Description | Value Range | Default |
    |--------|-------------|-------------|---------|
    |        |             |             |         |
BYPASS DIAL PLAN

- **Web Configuration**

  This option can be found under device web UI → Account → Accounts # → Call Settings → Bypass Dial Plan.

- **Functionality**

  Now the users can choose to bypass the dialplan on part of the function of the phone.

- **Pvalue**

<table>
<thead>
<tr>
<th>Pvalue</th>
<th>Description</th>
<th>Value Range</th>
<th>Default</th>
</tr>
</thead>
<tbody>
<tr>
<td>P2382/P2482/P2582</td>
<td>Bypass Dial Plan</td>
<td>Value = “Contact”</td>
<td>Mpk</td>
</tr>
<tr>
<td></td>
<td></td>
<td>Value = “Call History Incoming Call”</td>
<td></td>
</tr>
<tr>
<td></td>
<td></td>
<td>Value = “Call History Outgoing Call”</td>
<td></td>
</tr>
<tr>
<td></td>
<td></td>
<td>Value = “Dialing Page”</td>
<td></td>
</tr>
<tr>
<td></td>
<td></td>
<td>Value = “Mpk”</td>
<td></td>
</tr>
<tr>
<td></td>
<td></td>
<td>Value = “Api”</td>
<td></td>
</tr>
</tbody>
</table>

LIVE DIALPAD

- **Web Configuration**

  The option can be found under device web UI → Settings → Call Features → Live DialPad Expire Time.

- **Functionality**

  Now users can use Live DialPad to auto dial after a period of wait time. If you enable Live DialPad,
after dial in the numbers, the phone will wait for the time set in the expire time (in seconds) and dial out automatically after.

- **Pvalue**

<table>
<thead>
<tr>
<th>Pvalue</th>
<th>Description</th>
<th>Value Range</th>
<th>Default</th>
</tr>
</thead>
<tbody>
<tr>
<td>P8350</td>
<td>&lt;New Pvalue&gt; Bypass Dial Plan</td>
<td>Value = 0; No</td>
<td>No</td>
</tr>
<tr>
<td></td>
<td></td>
<td>Value = 1; Yes</td>
<td></td>
</tr>
<tr>
<td>P8351</td>
<td>&lt;New Pvalue&gt; Live DialPad Expire Time</td>
<td>2-15</td>
<td>5</td>
</tr>
</tbody>
</table>

**CID AND IMAGE FROM SIP NOTIFY**

- **Functionality**

  We have already supported the CID and Image previously. Now if phone receives NOTIFY message with type header: "application/cid-info+xml" and event header: "cid-info-update", phone will also take the message and update CID information and CID image.

**RANDOMIZED TR069 STARTUP**

- **Web Configuration**

  This option can be found under device web UI → Maintenance → TR-069 → Randomized TR069 Startup.

  ![Randomized TR069 Startup](…)

  - **Functionality**

    Now users can configure phone to randomize the TR069 update by enabling this feature. Random time will happen between 1-3600 seconds after bootup.

  - **Pvalue**

<table>
<thead>
<tr>
<th>Pvalue</th>
<th>Description</th>
<th>Value Range</th>
<th>Default</th>
</tr>
</thead>
<tbody>
<tr>
<td>P28116</td>
<td>&lt;New Pvalue&gt; Randomized TR069 Startup</td>
<td>Value = 0; Disabled</td>
<td>Disabled</td>
</tr>
<tr>
<td></td>
<td></td>
<td>Value = 1; Enabled</td>
<td></td>
</tr>
</tbody>
</table>

**OFFHOOK AUTO DIAL DELAY**

- **Web Configuration**

  This option can be found under device web UI → Settings → Call Features → Off-hook Auto Dial Delay.
Off-hook Auto Dial Delay

- **Functionality**

  Now if users set Off hook auto dial number, they can add a delay on it. Then the phone will wait for the time set before it auto dials out.

- **Pvalue**

<table>
<thead>
<tr>
<th>Pvalue</th>
<th>Description</th>
<th>Value Range</th>
<th>Default</th>
</tr>
</thead>
<tbody>
<tr>
<td>P8388</td>
<td>&lt;New Pvalue&gt; Off-hook Auto Dial Delay</td>
<td>0-30</td>
<td>0</td>
</tr>
</tbody>
</table>

**AUTO ANSWER DELAY**

- **Web Configuration**

  This option can be found under device web UI→ Settings → Call Features → Auto Answer Delay.

Auto Answer Delay

- **Functionality**

  Now if users set auto answer, they can also set a delay time before the phone is picked up. This way users can have more time to prepare for their incoming calls.

- **Pvalue**

<table>
<thead>
<tr>
<th>Pvalue</th>
<th>Description</th>
<th>Value Range</th>
<th>Default</th>
</tr>
</thead>
<tbody>
<tr>
<td>P8486</td>
<td>&lt;New Pvalue&gt; Auto Answer Delay</td>
<td>1-10</td>
<td>0</td>
</tr>
</tbody>
</table>

**MORE MATCHING RULES SUPPORTED**

- **Web Configuration**

  This option can be found under device web UI→ Accounts → Account # → Call Settings.
• Functionality

Now instead of supporting three matching rules ringtone settings, users can have up to ten matching incoming caller ID

• Pvalue

```
<table>
<thead>
<tr>
<th>Pvalue</th>
<th>Description</th>
<th>Value Range</th>
<th>Default</th>
</tr>
</thead>
<tbody>
<tr>
<td>Account 1-3:</td>
<td>&lt;New Pvalue&gt; Match Incoming Caller ID 4</td>
<td>String</td>
<td>N/A</td>
</tr>
<tr>
<td>P6716</td>
<td></td>
<td></td>
<td></td>
</tr>
<tr>
<td>P6722</td>
<td></td>
<td></td>
<td></td>
</tr>
<tr>
<td>P6728</td>
<td></td>
<td></td>
<td></td>
</tr>
<tr>
<td>Account 1-3:</td>
<td>&lt;New Pvalue&gt; Match Incoming Caller ID 5</td>
<td>String</td>
<td>N/A</td>
</tr>
<tr>
<td>P6718</td>
<td></td>
<td></td>
<td></td>
</tr>
<tr>
<td>P6724</td>
<td></td>
<td></td>
<td></td>
</tr>
<tr>
<td>P6730</td>
<td></td>
<td></td>
<td></td>
</tr>
<tr>
<td>Account 1-3:</td>
<td>&lt;New Pvalue&gt; Match Incoming Caller ID 6</td>
<td>String</td>
<td>N/A</td>
</tr>
<tr>
<td>P6720</td>
<td></td>
<td></td>
<td></td>
</tr>
<tr>
<td>P6726</td>
<td></td>
<td></td>
<td></td>
</tr>
<tr>
<td>P6732</td>
<td></td>
<td></td>
<td></td>
</tr>
<tr>
<td>Account 1-3:</td>
<td>&lt;New Pvalue&gt; Match Incoming Caller ID 7</td>
<td>String</td>
<td>N/A</td>
</tr>
<tr>
<td>P26064</td>
<td></td>
<td></td>
<td></td>
</tr>
<tr>
<td>P26164</td>
<td></td>
<td></td>
<td></td>
</tr>
<tr>
<td>P26264</td>
<td></td>
<td></td>
<td></td>
</tr>
</tbody>
</table>
```
<table>
<thead>
<tr>
<th>Account 1-3: P26066 P26166 P26266</th>
<th>&lt;New Pvalue&gt;</th>
<th>String</th>
<th>N/A</th>
</tr>
</thead>
<tbody>
<tr>
<td>Match Incoming Caller ID 8</td>
<td></td>
<td></td>
<td></td>
</tr>
</tbody>
</table>

<table>
<thead>
<tr>
<th>Account 1-3: P26068 P26168 P26268</th>
<th>&lt;New Pvalue&gt;</th>
<th>String</th>
<th>N/A</th>
</tr>
</thead>
<tbody>
<tr>
<td>Match Incoming Caller ID 9</td>
<td></td>
<td></td>
<td></td>
</tr>
</tbody>
</table>

<table>
<thead>
<tr>
<th>Account 1-3: P26096 P26196 P26296</th>
<th>&lt;New Pvalue&gt;</th>
<th>String</th>
<th>N/A</th>
</tr>
</thead>
<tbody>
<tr>
<td>Match Incoming Caller ID 10</td>
<td></td>
<td></td>
<td></td>
</tr>
</tbody>
</table>

<table>
<thead>
<tr>
<th>Pvalue</th>
<th>Description</th>
<th>Value Range</th>
<th>Default</th>
</tr>
</thead>
<tbody>
<tr>
<td>Account 1-3: P6717 P6723 P6729</td>
<td>&lt;New Pvalue&gt; Match Rule 4 Ringtone</td>
<td>Value= -1 No Ringtone; Value= 0 system ring tone; Value= 1 custom ring tone 1; Value= 2 custom ring tone 2; Value= 3 custom ring tone 3; Value= 4 silent; Value= 5 Default Ringtone.</td>
<td>Default Ringtone</td>
</tr>
<tr>
<td>Account 1-3: P6719 P6725 P6731</td>
<td>&lt;New Pvalue&gt; Match Rule 5 Ringtone</td>
<td>Value= -1 No Ringtone; Value= 0 system ring tone; Value= 1 custom ring tone 1; Value= 2 custom ring tone 2; Value= 3 custom ring tone 3; Value= 4 silent; Value= 5 Default Ringtone.</td>
<td>Default Ringtone</td>
</tr>
<tr>
<td>Account 1-3: P6721 P6727 P6733</td>
<td>&lt;New Pvalue&gt; Match Rule 6 Ringtone</td>
<td>Value= -1 No Ringtone; Value= 0 system ring tone; Value= 1 custom ring tone 1; Value= 2 custom ring tone 2; Value= 3 custom ring tone 3; Value= 4 silent; Value= 5 Default Ringtone.</td>
<td>Default Ringtone</td>
</tr>
<tr>
<td>Account 1-3: P26065 P26165 P26265</td>
<td>&lt;New Pvalue&gt; Match Rule 7 Ringtone</td>
<td>Value: -1 No Ringtone; Value: 0 system ring tone; Value: 1 custom ring tone 1; Value: 2 custom ring tone 2; Value: 3 custom ring tone 3; Value: 4 silent; Value: 5 Default Ringtone.</td>
<td>Default Ringtone</td>
</tr>
<tr>
<td>-----------------------------------</td>
<td>------------------------------------</td>
<td>----------------------------------------------------------------------------------</td>
<td>------------------</td>
</tr>
<tr>
<td>Account 1-3: P26067 P26167 P26267</td>
<td>&lt;New Pvalue&gt; Match Rule 8 Ringtone</td>
<td>Value: -1 No Ringtone; Value: 0 system ring tone; Value: 1 custom ring tone 1; Value: 2 custom ring tone 2; Value: 3 custom ring tone 3; Value: 4 silent; Value: 5 Default Ringtone.</td>
<td>Default Ringtone</td>
</tr>
<tr>
<td>Account 1-3: P26069 P26169 P26269</td>
<td>&lt;New Pvalue&gt; Match Rule 9 Ringtone</td>
<td>Value: -1 No Ringtone; Value: 0 system ring tone; Value: 1 custom ring tone 1; Value: 2 custom ring tone 2; Value: 3 custom ring tone 3; Value: 4 silent; Value: 5 Default Ringtone.</td>
<td>Default Ringtone</td>
</tr>
<tr>
<td>Account 1-3: P26097 P26197 P26297</td>
<td>&lt;New Pvalue&gt; Match Rule 10 Ringtone</td>
<td>Value: -1 No Ringtone; Value: 0 system ring tone; Value: 1 custom ring tone 1; Value: 2 custom ring tone 2; Value: 3 custom ring tone 3; Value: 4 silent; Value: 5 Default Ringtone.</td>
<td>Default Ringtone</td>
</tr>
</tbody>
</table>

**ENABLE ENHANCED ACOUSTIC ECHO CANCELLER**

- **Web Configuration**

  This option can be found under device web UI → Settings → Call Features → Enable Enhanced Acoustic Echo Canceller.
• **Functionality**

Now user is able to choose to enable or disable the echo canceller on the phone in speaker mode.

• **Pvalue**

<table>
<thead>
<tr>
<th>Pvalue</th>
<th>Description</th>
<th>Value Range</th>
<th>Default</th>
</tr>
</thead>
<tbody>
<tr>
<td>P22175</td>
<td>Enable Enhanced Acoustic Echo C canceller</td>
<td>Value = 0; No</td>
<td>Yes</td>
</tr>
<tr>
<td></td>
<td></td>
<td>Value = 1; Yes</td>
<td></td>
</tr>
</tbody>
</table>

**DOWNLOAD CERTIFICATE FILES DURING PROVISION**

• **Functionality**

Now if users enter server URL in the Pvalue listed below, phone will download the certificate file that’s on the server and commit into phone.

Those Pvalues are:
- 279: SIP TLS Certificate
- 280: SIP TLS Private Key
- 8433: Trusted CA Certificates 1
- 8434: Trusted CA Certificates 2
- 8435: Trusted CA Certificates 3
- 8436: Trusted CA Certificates 4
- 8437: Trusted CA Certificates 5
- 8438: Trusted CA Certificates 6
- 9902: OpenVPN CA
- 9903: OpenVPN Certificate
- 9904: OpenVPN Client key

**ON HOLD REMINDER TONE**

• **Web Configuration**

This option can be found under device web UI → Accounts → Account # → Call Settings → On Hold Reminder Tone.

- **Functionality**

Now the user can hear 3 beeps as hold reminder, the first on hold reminder occurs at 1 min and the subsequent on hold reminder occurs every 30 sec.

• **New Pvalue**
<table>
<thead>
<tr>
<th>Pvalue</th>
<th>Description</th>
<th>Value Range</th>
<th>Default</th>
</tr>
</thead>
<tbody>
<tr>
<td>P26006/P26106/P26206</td>
<td>&lt;New Pvalue&gt; On Hold Reminder Tone</td>
<td>Value = 0; Disabled</td>
<td>Disabled</td>
</tr>
</tbody>
</table>
FIRMWARE VERSION 1.0.4.106

PRODUCT NAME
GXP1610, GXP1615, GXP1620, GXP1625, GXP1628 and GXP1630

DATE
03/08/2018

FIRMWARE FILE INFORMATION
- Firmware file name: gxp1600fw.bin
  MD: d1d9c5aa986bc678a13b64155be07b29

ENHANCEMENT
- Enhanced syslog to run on other ports instead of default port [SYSLOG ENHANCEMENT]
- Enhanced intercom options (including Intercom Mute, Intercom Tone) [INTERCOM ENHANCEMENT]
- Added auto provision starts when certain p-values are changed [PVALUES THAT TRIGGER AUTO PROVISION]
- Added attempt to download config files [ATTEMPT TO DOWNLOAD CONFIG FILE AGAIN]
- Enhanced pressing speaker key takes 10-15 seconds to respond from idle to call screen after reboot

BUG FIX
- Fixed softkey button doesn’t work after logged in hoteling feature with Broadsoft
- Fixed 3CX phone provision with SIP NOTIFY Event header "check-sync;reboot=false" causes phone rebooting
- Fixed Phone leaves the stuck call in call park scenario with 3CX
- Fixed Phone does not send DND status back to PBX after 200OK for 3CX
- Remove MAC address in the "User-Agent" header of the SIP Register messages for 3CX
- Fixed Phone wrong sorting when using English and Czech phonebook contacts
- Fixed Phone does not remain the voice vlan when LLDP is set to voice vlan and voice-signaling vlan
- Fixed Phone is frozen after uploading wrong phone book

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

SYSLOG ENHANCEMENT
- Web Configuration
  This option can be found under device web UI→ Maintenance → Syslog → Syslog Server.
• **Functionality**

By adding port number to the syslog server field, the phone will send syslog to the corresponding port of that IP.

• **Pvalue**

<table>
<thead>
<tr>
<th>Pvalue</th>
<th>Description</th>
<th>Value Range</th>
<th>Default</th>
</tr>
</thead>
<tbody>
<tr>
<td>P207</td>
<td>Syslog Server</td>
<td>String</td>
<td>N/A</td>
</tr>
</tbody>
</table>

## INTERCOM ENHANCEMENT

• **Web Configuration**

There’re two new options for Intercom feature. These options can be found under device web UI→ Accounts → Account # → Intercom.

- **Mute on answer Intercom call**
  - Options: No, Yes

- **Play warning tone for Auto Answer Intercom**
  - Options: No, Yes

• **Functionality**

Now if phone receives intercom calls, and "Mute on answer intercom call" is enabled, when intercom calls are auto answered, phone will mute itself to increase privacy.

If “Play warning tone for auto answer Intercom” is enabled, when enabling auto answer, phone will play a warning tone to remind user that the intercom has been connected.

• **New Pvalue**

<table>
<thead>
<tr>
<th>Pvalue</th>
<th>Description</th>
<th>Value Range</th>
<th>Default</th>
</tr>
</thead>
<tbody>
<tr>
<td>P26071 (Account1)</td>
<td>&lt;New Pvalue&gt; Mute on answer Intercom call</td>
<td>Value = 0; No Value = 1; Yes</td>
<td>No</td>
</tr>
<tr>
<td>P26171 (Account 2)</td>
<td></td>
<td></td>
<td></td>
</tr>
<tr>
<td>P26271 (Account 3)</td>
<td></td>
<td></td>
<td></td>
</tr>
<tr>
<td>P26072 (Account1)</td>
<td>&lt;New Pvalue&gt; Play warning tone for Auto Answer Intercom</td>
<td>Value = 0; No Value = 1; Yes</td>
<td>No</td>
</tr>
<tr>
<td>P26172 (Account 2)</td>
<td></td>
<td></td>
<td></td>
</tr>
<tr>
<td>P26272 (Account 3)</td>
<td></td>
<td></td>
<td></td>
</tr>
</tbody>
</table>
PVALUES THAT TRIGGER AUTO PROVISION

• Functionality

If the p-values listed below are changed while managing configuration on web UI or LCD, the provision process will be triggered:

* 192 -- Firmware upgrade server
* 232 -- Firmware prefix
* 233 -- Firmware postfix
* 6767 -- Firmware Upgrade Via
* 6768 -- Firmware HTTP/HTTPS Username
* 6769 -- Firmware HTTP/HTTPS Password
* 237 -- Config upgrade Server
* 212 -- Config upgrade via
* 234 -- Config prefix
* 235 -- Config postfix
* 1360 -- Config HTTP/HTTPS username
* 1361 -- Config HTTP/HTTPS password

ATTEMPT TO DOWNLOAD CONFIG FILE AGAIN

• Functionality

When doing provision on the phone, if your first config file contains p-values listed below, phone will try to download the potential second cfg.xml file and apply the second file without rebooting. Maximum 3 extra attempts. Those P-values are:

*212 -- Config upgrade via
*234 -- Config prefix
*235 -- Config postfix
*237 -- Config upgrade Server
*240 -- Authenticate Config File
*1359 -- XML Config File Password
*8463 -- Validate Server Certificate
*8467 -- Download and process ALL Available Config Files
*20713 -- Always authenticate before challenge
*22030 -- Validate Hostname in Certificate
FIRMWARE VERSION 1.0.4.100

PRODUCT NAME
GXP1610, GXP1615, GXP1620, GXP1625, GXP1628 and GXP1630

DATE
02/16/2018

ENHANCEMENT
• Added support for VPK share line to display description value [VPK SHARED LINE DESCRIPTION]
• Added option to enable BLF or speed dial VPK/MPK to perform blind transfer during active call [ENABLE TRANSFER VIA NON-TRANSFER MPK]
• Allowed LCD backlight to be always on or always off [SUPPORT TO ALLOW LCD BACKLIGHT ALWAYS ON]
• Added “Let's Encrypt” Validation for 3CX
• Added support to automatically log in web UI from server interface for 3CX [AUTOMATICALLY LOG IN FROM 3CX SERVER INTERFACE]
• Added uaCSTA support for 3CX [uaCSTA SUPPORT FOR 3CX]
• Added monitored call park MPK option and supported to be provisioned by 3CX
• Added Use SBC configuration for 3CX [USE SBC WITH 3CX]
• Added support for HTTPS server based on TLSv1.2
• Added option to disable user web access [USER WEB ACCESS]
• Enabled strong password for admin/user password [TEST PASSWORD STRENGTH]
• Force admin to change default password upon first time login
• Added option to include MAC address in the SIP User-Agent [USE MAC HEADER]
• Added Server Validation [SERVER VALIDATION]
• Added support for Cisco Discovery Protocol (CDP) [SUPPORT FOR CISCO DISCOVERY PROTOCOL (CDP)]
• Added option to choose either to override or append phonebook groups when a new XML file is uploaded [IMPORT GROUP METHOD]

BUG FIX
• Fixed Device carries two user-agent headers
• Fixed Device User-Agent carries error information
• Fixed Device gets frozen after uploading wrong phone book
• Fixed Device uploads group number abnormally
• Fixed Device gets frozen when uploading a phonebook with maximum contacts
• Fixed Import Phonebook Method is shown as XXXX
• Fixed Registration failure after setting up validate certificate chain to “Yes”
• Fixed Device takes too long to do failover from primary SIP server to secondary SIP server when SIP/TCP is set on the phone
• Fixed Device time zone is not updated after re-provisioning
• Updated Mozilla certificate files
• Updated Grandstream product certificate
• Fixed Device not sending RTP nor decoding RTP stream
• Fixed Recovery core replaces normal core issue
• Fixed Device does not provision if new configuration is the same as previous received configuration
• Fixed Server certificate validation not working
• Fixed Device cannot parse 3CX generated phonebook.xml
• Fixed Device crashes after answering the queue call with 3CX
• Fixed The number of contacts is not limited
• Fixed Device phonebook feature does not work correctly
• Fixed Device randomly crashes when switching fast between the lines
• Fixed Device always sends SIP requests to OBP port 5060 when configured as a different port
• Fixed Device does not send unregister after factory reset
• Fixed Device recognizes * as asterisk for 3CX CTI
• Fixed Device SNMP Trap Version works abnormally after modifying version
• Fixed User CA File Path
• Set Config upgrade via HTTPS & Firmware via HTTP as default value
• Fixed Wrong sorting issue when having English and Czech phonebook contacts
• Fixed There is no DHCP INFORM when ACS URL is not set
• Fixed Device cannot handle TCP packets that contain multiple SIP messages
• Fixed Modifying SIP SUBSCRIBE expiration doesn't trigger new registration/subscription

**NEW FEATURES OVERVIEW**
This section lists major new features and describes how to use it from the user’s point of view.

**VPK SHARED LINE DESCRIPTION**

- **Functionality**

If the VPK Description is set, it will show the description on the LCD screen. If the Description is left empty, Default value will be Account name.

**ENABLE TRANSFER VIA NON-TRANSFER MPK**

- **Web Configuration**

This option can be found under device web UI → Settings → Programmable Keys → Multi-Purpose Keys Settings → Enable transfer via non-Transfer MPK.
• Functionality

Now users can setup their VPK or MPK as BLF or Speed Dial, and use such VPK to perform transfer in active calls.

• New Pvalue

<table>
<thead>
<tr>
<th>Pvalue</th>
<th>Description</th>
<th>Value Range</th>
<th>Default</th>
</tr>
</thead>
<tbody>
<tr>
<td>P22162</td>
<td>&lt;New Pvalue&gt;</td>
<td>Value = 0; No Value = 1; Yes</td>
<td>No</td>
</tr>
</tbody>
</table>

SUPPORT TO ALLOW LCD BACKLIGHT ALWAYS ON

• Web Configuration

This option can be found under device web UI → Settings → Preferences → LCD Display → Active Back Light Timeout.

![Active Backlight Timeout](0)

• Functionality

When Active Backlight Timeout is set to 0, the backlight will be constantly on.

• New Pvalue

<table>
<thead>
<tr>
<th>Pvalue</th>
<th>Description</th>
<th>Value Range</th>
<th>Default</th>
</tr>
</thead>
<tbody>
<tr>
<td>P8356</td>
<td>Active Backlight Timeout</td>
<td>0(Always on), 1-90 (mins)</td>
<td>1</td>
</tr>
</tbody>
</table>

AUTOMATICALLY LOG IN FROM 3CX SERVER INTERFACE

• Functionality

Now users can log on to the phone’s web UI from 3CX server by pressing the Phone UI with the correct provision and correct template.

uaCSTA SUPPORT FOR 3CX

• Web Configuration

This option can be found under device web UI → Network → Remote Control → CSTA Control.
• **Functionality**

Now the users can use 3CX server with the CSTA enabled and have the following functions supported:

- Support for CSTA session establish through SIP INVITE
- Support for CSTA MonitorStart request through SIP INFO
- Support for CSTA MakeCall request through SIP INFO
- Support for CSTA AnswerCall request through SIP INFO
- Support to send CSTA DeliveredEvent request through SIP INFO

• **New Pvalue**

<table>
<thead>
<tr>
<th>Pvalue</th>
<th>Description</th>
<th>Value Range</th>
<th>Default</th>
</tr>
</thead>
<tbody>
<tr>
<td>P32053</td>
<td>&lt;New Pvalue&gt; CSTA Control</td>
<td>Value = 0; Disabled Value = 1; Enabled</td>
<td>Disable</td>
</tr>
</tbody>
</table>

**USER WEB ACCESS**

• **Web Configuration**

This option can be found under device web UI → Maintenance → Security Settings → Security → Enable User Web Access.

- **Functionality**

When this option is disabled, the user account will no longer be able to log on to web GUI. This allow admin to be fully control for the phone.

• **New Pvalue**

<table>
<thead>
<tr>
<th>Pvalue</th>
<th>Description</th>
<th>Value Range</th>
<th>Default</th>
</tr>
</thead>
<tbody>
<tr>
<td>P8469</td>
<td>&lt;New Pvalue&gt; Enable User Web Access</td>
<td>Value = 0; Disabled Value = 1; Enabled</td>
<td>Disabled</td>
</tr>
</tbody>
</table>

**TEST PASSWORD STRENGTH**

• **Web Configuration**

This option can be found under device web UI → Settings → General Settings → Test Password Strength.
- **Functionality**

  When choose Yes, users are able to set their password contains at least 3 out of these four categories:
  
  1. Numerics (0-9)
  2. Capital Letters (A-Z)
  3. Lower Case Letters (a-z)
  4. Special symbols: Eg: !, @, $, %, ^, &, *, (, ), etc.

  Also that the minimum characters should not be less than 9. This allows more security for the user/admin.

- **New Pvalue**

<table>
<thead>
<tr>
<th>Pvalue</th>
<th>Description</th>
<th>Value Range</th>
<th>Default</th>
</tr>
</thead>
<tbody>
<tr>
<td>P8468</td>
<td>Test Password Strength</td>
<td>Value = 0; No</td>
<td>No</td>
</tr>
<tr>
<td></td>
<td></td>
<td>Value = 1; Yes</td>
<td></td>
</tr>
</tbody>
</table>

**USE MAC HEADER**

- **Web Configuration**

  This option can be found under device web UI → Accounts → Account # → SIP Settings → Custom SIP Headers.

  ![Use MAC Header](No Yes)

- **Functionality**

  Now instead of showing in regular MAC field, we have included the MAC address into the User-Agent Field.

- **New Pvalue**

<table>
<thead>
<tr>
<th>Pvalue</th>
<th>Description</th>
<th>Value Range</th>
<th>Default</th>
</tr>
</thead>
<tbody>
<tr>
<td>P26061</td>
<td>Use MAC Header</td>
<td>Value = 0; User Name</td>
<td>User Name</td>
</tr>
<tr>
<td></td>
<td></td>
<td>Value = 1; User ID</td>
<td></td>
</tr>
</tbody>
</table>

**SERVER VALIDATION**

- **Web Configuration**

  This option can be found under device web UI → Maintenance → Security Settings → Security → Validate Server Certificate.
• **Functionality**

When enabling this feature, the phone will then validate the server’s certificate. If the server that our phone tries to register on is not on our list, it will not allow server to access our phone.

• **New Pvalue**

<table>
<thead>
<tr>
<th>Pvalue</th>
<th>Description</th>
<th>Value Range</th>
<th>Default</th>
</tr>
</thead>
<tbody>
<tr>
<td>P8463</td>
<td>&lt;New Pvalue&gt; Validate Server Certificates</td>
<td>Value = 0; No Value = 1; Yes</td>
<td>Yes</td>
</tr>
</tbody>
</table>

**SUPPORT FOR CISCO DISCOVERY PROTOCOL (CDP)**

• **Web Configuration**

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

<table>
<thead>
<tr>
<th>Enable CDP</th>
<th>Enabled</th>
<th>Disabled</th>
</tr>
</thead>
</table>

• **Functionality**

Allow user to enable/disable CDP feature which will allow the phone to receive CDP packets from the uplink. When CDP feature is enabled, make sure VLAN is also configured accordingly. It will need reboot to take effect for both enabling and disabling this feature.

• **New Pvalue**

<table>
<thead>
<tr>
<th>Pvalue</th>
<th>Description</th>
<th>Value Range</th>
<th>Default</th>
</tr>
</thead>
<tbody>
<tr>
<td>P22119</td>
<td>&lt;New Pvalue&gt; Enable CDP</td>
<td>Value = 0; Enabled Value = 1; Disabled</td>
<td>Enabled</td>
</tr>
</tbody>
</table>

**IMPORT GROUP METHOD**

• **Web Configuration**

This option can be found under device web UI → Phonebook → Phonebook management → Import Group Method.

<table>
<thead>
<tr>
<th>Import Group Method</th>
<th>Replace</th>
<th>Append</th>
</tr>
</thead>
</table>

• **Functionality**

Now when user imports a new phonebook, they can choose whether to replace the old phonebook groups or combine old phonebook and new phonebook together.
• New Pvalue

<table>
<thead>
<tr>
<th>Pvalue</th>
<th>Description</th>
<th>Value Range</th>
<th>Default</th>
</tr>
</thead>
<tbody>
<tr>
<td>P8462</td>
<td>&lt;New Pvalue&gt; Import Group Method</td>
<td>Value = 0; Replace Value = 1; Append</td>
<td>Replace</td>
</tr>
</tbody>
</table>

USE SBC WITH 3CX

• Web Configuration

This option can be found under device web UI → Accounts → Account # → Network Settings → Use SBC.

![Use SBC](No Yes)

• Functionality

Now if users want to work under SBC associated with 3CX, they should enable this feature to have better communication with the server.

• New Pvalue

<table>
<thead>
<tr>
<th>Pvalue</th>
<th>Description</th>
<th>Value Range</th>
<th>Default</th>
</tr>
</thead>
<tbody>
<tr>
<td>P26090</td>
<td>&lt;New Pvalue&gt; Use SBC</td>
<td>Value = 0; No Value = 1; Yes</td>
<td>No</td>
</tr>
<tr>
<td>P26190</td>
<td></td>
<td></td>
<td></td>
</tr>
<tr>
<td>P26290</td>
<td></td>
<td></td>
<td></td>
</tr>
</tbody>
</table>
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
- Fixed phone 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.

  ![Account Display Configuration](image)

• 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

<table>
<thead>
<tr>
<th>Pvalue</th>
<th>Description</th>
<th>Value Range</th>
<th>Default</th>
</tr>
</thead>
<tbody>
<tr>
<td>P2380 (Account 1)</td>
<td>Account Display</td>
<td>Value = 0; User Name</td>
<td>User Name</td>
</tr>
<tr>
<td>P2480 (Account 2)</td>
<td></td>
<td>Value = 1; User ID</td>
<td></td>
</tr>
<tr>
<td>P2580 (Account 3)</td>
<td></td>
<td></td>
<td></td>
</tr>
</tbody>
</table>

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.
LED Control

**Functionality**
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**

<table>
<thead>
<tr>
<th>Pvalue</th>
<th>Description</th>
<th>Value Range</th>
<th>Default</th>
</tr>
</thead>
<tbody>
<tr>
<td>P8371</td>
<td>Disable VM/MSG Power Light Flash</td>
<td>Value = 0; No Value = 1; Yes</td>
<td>No</td>
</tr>
</tbody>
</table>

SUPPORT ACCESS SSH USING AUTHENTICATION KEYS

**Web Configuration**
The upload option for “SSH Public Key” can be found under device web UI->Maintenance->Security.

**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.
• **Functionality**
  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**

<table>
<thead>
<tr>
<th>Pvalue</th>
<th>Description</th>
<th>Value Range</th>
<th>Default</th>
</tr>
</thead>
<tbody>
<tr>
<td>P22102</td>
<td>Ring For Call Waiting</td>
<td>Value = 0; No</td>
<td>No</td>
</tr>
<tr>
<td></td>
<td></td>
<td>Value = 1; Yes</td>
<td></td>
</tr>
</tbody>
</table>

**SUPPORT CALL FORWARD CONFIGURATION ON WEB UI**

• **Web Configuration**
  Call forwarding configurations can be found under device web UI→Accounts→Account x→Feature Codes.

**Feature Codes**

**Call Forward Unconditionally (All)**
- **Target**

**Call Forward Busy**
- **Target**

**Call Forward Delayed (No Answer)**
- **Target**
- **Delayed Call Forward Wait Time**

• **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**

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

**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](Attended Transfer Mode.png)

  - **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**

<table>
<thead>
<tr>
<th>Pvalue</th>
<th>Description</th>
<th>Value Range</th>
<th>Default</th>
</tr>
</thead>
<tbody>
<tr>
<td>P1376</td>
<td>Attended Transfer Mode</td>
<td>Value = 0; Static</td>
<td>Static</td>
</tr>
<tr>
<td></td>
<td></td>
<td>Value = 1; Dynamic</td>
<td></td>
</tr>
</tbody>
</table>

**SUPPORT BLF SERVER CONFIGURATION**

- **Web Configuration**
  The “BLF Server” option can be found under device web UI->Accounts->Account X->General Settings.
• **Functionality**
  When “BLF Server” is configured, the SUBSCRIBE request will be sent to this address for BLF.

**# BLF Server**

• **New Pvalue**

<table>
<thead>
<tr>
<th>Pvalue</th>
<th>Description</th>
<th>Value Range</th>
<th>Default</th>
</tr>
</thead>
<tbody>
<tr>
<td>P2375 (Account 1)</td>
<td>BLF Server</td>
<td>String</td>
<td>None</td>
</tr>
<tr>
<td>P2475 (Account 2)</td>
<td>BLF Server</td>
<td>String</td>
<td>None</td>
</tr>
<tr>
<td>P2575 (Account 3)</td>
<td>BLF Server</td>
<td>String</td>
<td>None</td>
</tr>
</tbody>
</table>

**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.

```
Enable Automatic Redial

Automatic Redial Times

Automatic Redial Interval
```

• **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

<table>
<thead>
<tr>
<th>Pvalue</th>
<th>Description</th>
<th>Value Range</th>
<th>Default</th>
</tr>
</thead>
<tbody>
<tr>
<td>P8353</td>
<td>Enable Automatic Redial</td>
<td>Value = 0; No</td>
<td>No</td>
</tr>
<tr>
<td></td>
<td></td>
<td>Value = 1; Yes</td>
<td></td>
</tr>
<tr>
<td>P8354</td>
<td>Automatic Redial Times</td>
<td>Number 1 to 100</td>
<td>10</td>
</tr>
<tr>
<td>P8355</td>
<td>Automatic Redial Interval</td>
<td>Number 1 to 360</td>
<td>20</td>
</tr>
</tbody>
</table>

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.
• **Functionality**
  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](image)

  **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**

<table>
<thead>
<tr>
<th>Pvalue</th>
<th>Description</th>
<th>Value Range</th>
<th>Default</th>
</tr>
</thead>
<tbody>
<tr>
<td>P6763</td>
<td>Disable Hook Switch</td>
<td>Value = 0; No; Value = 1; Yes; Value = 2; For Answering Call</td>
<td>No</td>
</tr>
</tbody>
</table>

**SUPPORT NOTIFICATION TONE VOLUME**

• **Web Configuration**
  Notification tone volume can be adjusted under web UI->Settings->Ring Tone.

  ![Notification Tone Volume](image)

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

• **New Pvalue**

<table>
<thead>
<tr>
<th>Pvalue</th>
<th>Description</th>
<th>Value Range</th>
<th>Default</th>
</tr>
</thead>
<tbody>
<tr>
<td>P8399</td>
<td>Notification Tone Volume</td>
<td>Value = 0 to 7</td>
<td>4</td>
</tr>
</tbody>
</table>
SUPPORT SNMP

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

SNMP Settings

<table>
<thead>
<tr>
<th>Pvalue</th>
<th>Description</th>
<th>Value Range</th>
<th>Default</th>
</tr>
</thead>
<tbody>
<tr>
<td>P21896</td>
<td>Enable SNMP</td>
<td>Value = 0; No</td>
<td>0</td>
</tr>
<tr>
<td></td>
<td></td>
<td>Value = 1; Yes</td>
<td></td>
</tr>
<tr>
<td>P21900</td>
<td>SNMP Trap Community</td>
<td>String</td>
<td></td>
</tr>
<tr>
<td>P21897</td>
<td>SNMP Trap IP</td>
<td>String</td>
<td></td>
</tr>
<tr>
<td>P21898</td>
<td>SNMP Trap Port</td>
<td>Number: 1025 to 65535</td>
<td>162</td>
</tr>
<tr>
<td>P21899</td>
<td>SNMP Trap Version</td>
<td>Value = 0; Version 1</td>
<td>Version 1</td>
</tr>
<tr>
<td></td>
<td></td>
<td>Value = 1; Version 2</td>
<td></td>
</tr>
<tr>
<td>P21901</td>
<td>SNMP Trap Interval</td>
<td>Number: 1 to 1440</td>
<td>5</td>
</tr>
</tbody>
</table>

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.
• **Functionality**
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**

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