XBLUE IP9g RevB (V2) Release Notes

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Revision History

Version	Release Date
2.12.2	2022.3.10
2.12.X	2022.8.19
2.12.15.1	2022.10.11
2.12.15.8	2022.12.16
2.12.16.17.1	2023.06.30

1) New Features

• Add Croatian language.

2) Optimization

- Optimize DNS query mode.
- Optimize the time to return to the Dsskey Home.
- Optimize the Call Record function by adding the field "Reason" (for this call).

3) Bug Fixes

- Fixed display font to allow fully displayed Softkey tag "Intercom".
- Fixed the issue that IPv4 IP acquisition failures are frequently displayed after connecting to Wi-Fi.
- Fixed the problem that the help information prompt of the Dsskey icon on the phone web page is incorrect.
- Fixed the problems with MAC being available at phone restart.
- Fixed the problem with phone sending STUN packets when the STUN is stopped on the server.
- Fix the problem with delay when answering with the handset.
- Fix the problem with abnormal certification verification.
- Fix the problem with restarting under unique conditions of Provision Config Priority.
- Fix the problem of LED light on the secondary screen will not change with the number of pages.

Release 2.12.15.8

1) New Features

None

2) Optimization

• Optimize the problem of French translation

3) Bug Fixes

None

Release 2.12.15.1 (First Release)

1) New Features

- Add Italian language.
- Add the "Mcast Listening Renew Time" feature.
- Support for sending Action URL via SIP info.
- Added a <=> rule to Dial Plan that replaces the number before the equal sign and inserts the number after the equal sign.
- Support user custom settings of SIP CallerID fields.
- Support the status of the DND using BLF under a specific server.

2) Optimization

- Support for modifying the corresponding ringtones for Bellcore-drX
- Optimize the ringtone files
- Optimize the function of voice message button. This feature solves the problem that
 regardless of whether a SIP account is configured or not, the phone can press the
 Voice Message button on the LCD to dial out the voice message number.

3) Bug Fix

- Repair the problem unable to adjust Ringing tone cannot be adjusted at the phone after the number is dialed in the speaker mode
- Fix the problem of non-responsive transfer failure
- Repair the VLAN-related issues
- Fix Alert Info special ringtones. This function refers to the situation where the phone still plays the ringtone below the line and does not play the ringtone set by the alert information when with the configured Alert-Info is received during an incoming call.
- Fix the problem that the Cisco switch cannot access the WEB page under special Circumstances The feature refers to Cisco's switch and the environment is divided into a voice VLAN. After obtaining the address, the phone was used normally, but after using it for a period of time, the phone registration could not be reached and the phone webpage could not be accessed. The actual display of the phone is that it has an IP address, and only by restarting the phone will it be restored to normal.
- Fix the LDAP Contact-related issues
- Fix the phone hands-free playback voice distortion problem
- Repair phone access to IPV6 network, web page shows unconnected problem
- Fix issues that Action URL was unable to report missed incoming calls
- Repair the configuration XML browser and reuse the user name and password of the SIP account, causing the URL to stop working

Release 2.12.X (Not released)

1) New Features

None

2) Optimization

• Optimize translation issues related to contacts.

3) Bug Fixes

None

Release 2.12.2 (Not released)

- 1) New Features
- 2) Optimization
- 3) Bug Fixes
 - In some cases, if you use G723 and G726 codecs to establish a call, the phone may have abnormal faults.