



Release Note

Software Version: 2.12.10

Models: i10S/i10SD/i10SV/i16S/i16SV/PA3/PA2S/i6x

Date: 2023.1.11

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1 Revise History

No.	Version	Release Time
1	2.4.4	2021.5.25
2	2.12.4	2022.8.2
3	2.12.2	2022.9.21
4	2.12.10	2023.1.11

2 Release 2.12.10

Models	Software Version	Comment
i10S/ i10SD/i10SV/ i16S/i16SV/PA3	2.12.10	
i61/i62/i63/i64	2.12.10	

2.1 New Features

1. Add Audio Delay Playback feature
2. Support custom setting of HTTP API password and user name

2.2 Optimization

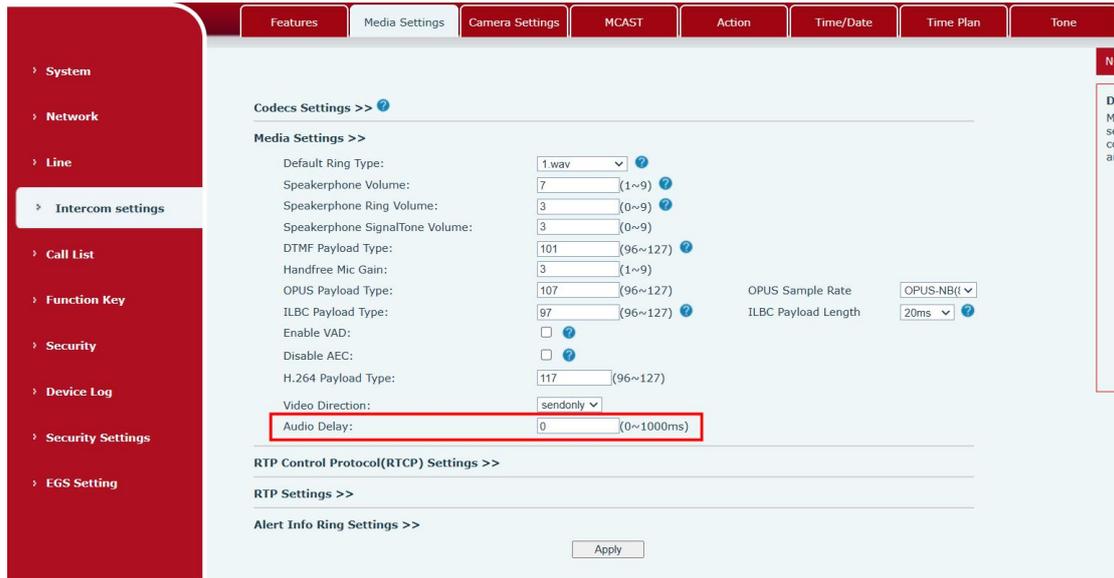
1. Add a transfer number to the password list. When the phone number can not get through, call the transfer number (only for Door Phone products)
2. Expand the total number of speed-dial button setting numbers to 8
3. When the device is set to static IP and is not connected to the network, the corresponding IP can still be announced

2.3 New Functions Description

1. Add Audio Delay Playback Feature

Function description: Add configuration to control the sound delay playback on the device to solve the problem of multiple PAs playing out of sync in the same space. Set the delay time in the voice setting, and users can adjust the delay time of device playback according to the actual situation.

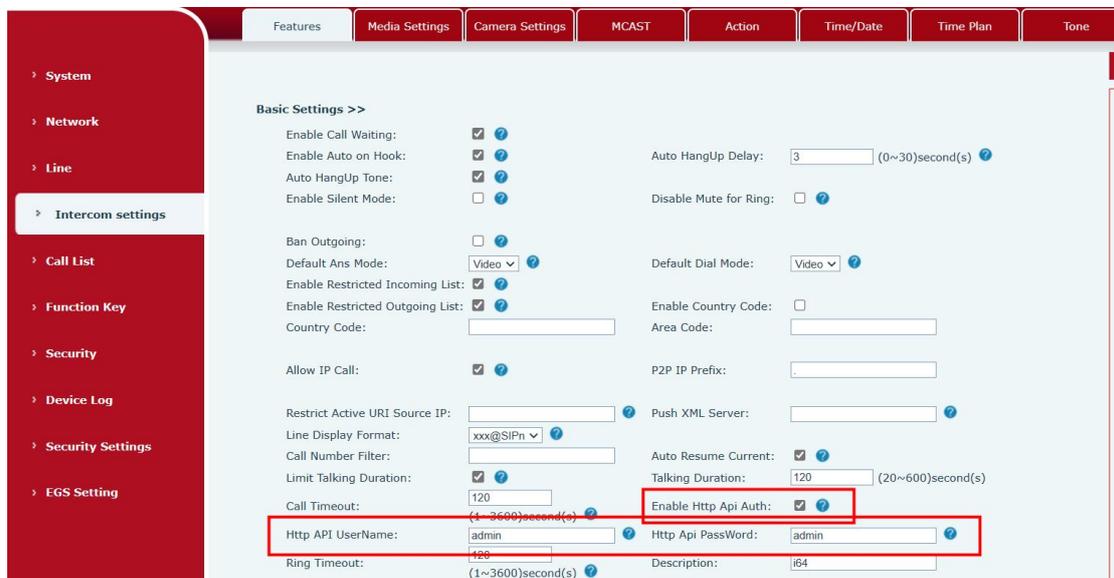
Setting method: On the website: [Intercom Settings] >> [Media Settings], set the Audio delay time.



2. Support custom setting of HTTP API password and user name

Function description: The device supports setting custom HTTP API usernames and passwords to increase the security level of the device.

Setting method: On the website [Intercom Settings] >> [Features], Set the "HTTP API user name" and "HTTP API password" configuration. The default user name and password are used for authentication, if the user has a custom user name and password, use the custom one for authentication, the default user name and password are admin, admin



2.4 Optimization Functions Description

1. Add a transfer number to the password list. When the phone number can not get through, call the transfer number

Function description: Enables location speed dialing to call a set number and, in the absence of a response, to call the set forwarding number

Setting method: On the web side, enter the [EGS Settings] >> [Password]>> [Add Password Rule]. Select the Add Password Rule module and fill in the number and the call forwarding number at the number.

The screenshot shows the 'Add Password Rule' configuration page in the Fanvil web interface. The page is divided into several sections:

- Import Password List:** Includes a 'Choose file:' field, a 'Select' button, and an 'Update' button.
- Add Password Rule:** Contains various configuration options:
 - Type: Local (dropdown)
 - Relay: Relay1 Relay2
 - Mode: Enable (dropdown)
 - Times: (input field)
 - Name: (input field)
 - Password: (input field)
 - Number: (input field)
 - Location: (input field)
 - CallForward: (input field, highlighted with a red box)
- Period:** Includes 'All Schedules' and 'Enable Schedules' sections with arrows for navigation.
- Buttons:** 'Add' and 'Modify' buttons are located at the bottom of the configuration section.
- Password List:** A table listing existing password rules. The table has columns: Index, Name, Type, Password, Relay, Mode, Number, Location, CallForward, Times, Period, Source, Issuing Date. The total count is 3.

3 Release 2.12.2

Models	Software Version	Comment
i10S/ i10SD/i10SV/ i16S/i16SV/PA3	2.12.2	
i61/i62/i63/i64	2.12.2	
PA2S	2.12.11	

3.1 New Features

3. Add the function of automatic hanging up after the intercom opens the door
4. Support the function of multicast local music through key
5. Add the function of regular multicast music
6. Local broadcast supports line out output (PA3)
7. The web page supports exporting captured pictures (only for Door Phone products)
8. Add location speed dialing function (only for Door Phone products)
9. Support the function of adding and deleting card search and password list through HTTP API (only for Door Phone products)
10. The web page supports displaying USB and TF card information
11. The function key supports setting multiple numbers and supports the sequential call function
12. Support 802.1x function
13. Support the configuration of DHCP host name
14. Add OpenVPN to support TLS-auth key
15. Support equipment Echo Test function
16. Support the function of sending offline broadcast through SIP message
17. Add guest user to support editing time plan

18. Add the function of custom uploading audio files and playing the specified audio files after the call and call play.

3.2 Optimization

1. Improve Multicast, Intercom and SIP calls, which can be inserted into each other
2. When using USB flash disk or SD card to play music, the call interrupts the music playing, and the music playing will resume automatically after the call ends.
3. Improve playing the door opening prompt tone in the call and send it to the opposite end (only for Door Phone products)
4. Turn off the startup prompt tone by default(PA3, PA2S)
5. Improve the webpage to support setting alert tone and busy tone
6. Improve the prompt tone of the remote DTMF trigger output port, and support user-defined settings
7. The device supports dynamic downloading of ringtone files through the URL of alert info
8. When the input port is used as a function key, the behavior in various states can be set
9. Improve the default enabling of dialing by "#" (i64)
10. Improve the ping function, and automatically return and display the results in case of failure
11. Support the device to push the configuration through HTTP API during the call
12. Improve the maximum value of talking duration to 12h
13. Support dynamic addition of function key values
14. Compatible with IPTech View platform

3.3 Bug Fix

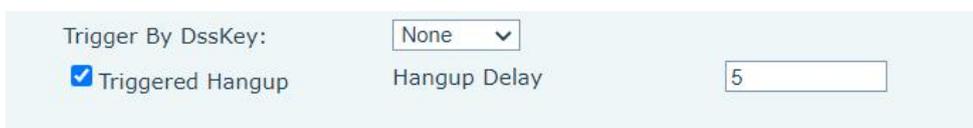
1. Fix the problem that audio cannot be sent to the opposite end when line in is used as the called call (PA3)
2. Fix the problem that the voice fails when the device is calling in some environments
3. Fix the problem that the device does not take effect after VLAN is enabled

4. Fix the problem that the device is repeatedly disconnected from L2TP VPN
5. Fix the problem that when playing timed audio via USB flash drive, the file name is Chinese audio and cannot be played
6. Fix the problem that the content played in the first 1s of multicast is incomplete when receiving multicast
7. Fix the problem that the device can't actively hang up the call after the call is interrupted in the middle of the broadcast
8. Fixed the problem that the call was automatically hung up when the PTT function was used and the key was pressed for 5S
9. Fix the problem of invalid volume adjustment in standby mode (PA3)
10. Fix the problem that there is no ring back when the device calls out in some environments
11. Fixed the problem that when connecting an external camera, the ip could not be called occasionally after using it for a period of time (PA2S)

3.4 New Functions Description

1. Add the function of automatic hanging up after the intercom opens the door

Function description: After the device triggers the output port by inputting DTMF, the device can automatically end the current call after the set on hook waiting time.



Trigger By DssKey:
 Triggered Hangup
 Hangup Delay

2. Support the function of multicast local music through key

Function description: By pressing the DSS key, you can play or pause local music files in a multicast manner.

Setting method: in the shortcut key setting, select "mcast paging" as the shortcut key type, set the value of 1 as the address of multicast, set the value of 2 as the path of local music file, select "remote only" or "remote&Local" as the media, and set the subtype to the corresponding encoding type to realize the function of local music playback.

"Remote Only" means that only local music is played at the remote end;

"Remote&Local" means that local music files are played at the same time in remote and local areas.

Function Key Settings >>

Key	Type	Name	Value	Subtype	Line	Media
DSS Key 1	MCAST Pagin		239.1.1.1:1378	G.711U	132@SIP1	Remote Only
			\\music\1.mp3			

3. Add the function of regular multicast music

Function description: Within the set timing period, the set audio file can be played in the group domain.

Setting method: Under the setting of scheduled audio playback, you can set the local scheduled playback, multicast scheduled playback, and local and multicast scheduled playback. Among them, when the multicast is selected to play regularly, the corresponding multicast address can be set to realize the multicast function of playing music in different situations; Select local scheduled playback, and the audio will be played automatically at the set scheduled time; Select local and multicast scheduled playback, and the devices on the local and corresponding multicast addresses will automatically play audio within the set timing period.

Time Plan:

Name:

Type: Timed play audio

Audio Path: Local

Audio Settings: 1.mp3

Play mode: Circle

Play Type: Mcast

Mcast Addr:

Codec: Local&Mcast

Repetition period: No repetition

Monthly: 1 2 3 4 5 6

4. Local broadcast supports line out output (PA3)

Function description: The set broadcast can be output and played in the form of line out in all, caller / callee / intercom / multicast and Media states. Among them, **All** ----- all audio is output from Line out, **Media** ----- all local audio is output from line out, and **Call (calling/called/intercom/multicast)** ----- when only calling/called/intercom/multicast is established in a call, audio files are output from line out.

5. The web page supports exporting captured pictures (only for Door Phone products)

Function description: When the storage mode is SD or Both, you can export the saved pictures in the SD card to the local, and view the captured pictures.

Snapshot

Snapshot Trigger Mode:

Snapshot By Input: Input1 Input2 Input3

Snapshot By State: Talking Ringing Calling

Snapshot By Motion Detection:

Snapshot Save: Server SD Both

Server Url:

Username: Password:

[Right Click here to Save Camera Photo](#)

6. The function key supports setting multiple numbers and supports the sequential call function

Function description: Each dsskey supports adding 5 numbers. When the first call is blocked, rejected or unanswered, the next number will be called in sequence.

Function Key Settings >>

Key	Type	Name	Value	Subtype	Line	Media
DSS Key 1	Memory Key		312	Speed Dial	132@SIP1	DEFAULT
			123			
			456			
			789			
			321			

7. Add the function of custom uploading audio files and playing the specified audio files after the call and call play

Function description : Support user-defined uploading of audio files. When talking with a specified number, the user can play the customized audio files to the opposite end. After the set number of plays is completed, the call will be automatically hung up. Or play customized audio files locally after the call ends.

Action Plan Add

Action: Play Audio

Number: Type: Connected

Direction: Both Line: 1011@SIP1

Audio File: open-failed.wav Play Times:

8. Add location speed dialing function (only for door phone products)

Function description: Set the location speed dial number. When calling the location speed dial (room number), you can call the corresponding host number to open the door remotely.

Add Password Rule

Type: Remote

Relay: Relay1 Relay2

Mode: Enable

Times:

Name:

Password:

Number:

Location:

Period: All Schedules Enable Schedules

3.5 Optimization Functions Description

1. Improve Multicast, Intercom and SIP calls, which can be inserted into each other

Function description: Set Multicast, Intercom, and SIP priorities. The smaller the number, the higher the priority. 1 is the highest priority. When there is a multi-way call, there is a high-priority call currently, and a low-priority call is in a waiting state; there is a low-priority call currently, and a high-priority call can preempt the low-priority call.

Priority: SIP Priority

Enable Page Priority: Multicast Priority

MCAST Listening

Priority: Intercom Priority:

Enable Page Priority: Mcast Listening Renew Time:

Enable Prio Chan:

Enable Emer Chan:

Index/Priority	Name	Host:port	Channel
1	<input type="text"/>	<input type="text"/>	0 <input type="text" value="0"/>
2	<input type="text"/>	<input type="text"/>	0 <input type="text" value="0"/>

2. Turn off the startup prompt tone by default

Function description: If this option is checked, there will be no startup prompt sound when the equipment is powered on; Otherwise, if skip boot tone is unchecked, it is considered that there is a prompt tone when the device is powered on. **Only PA2S and PA3 are off by default.**

Tone Settings >>

<p>Enable Holding Tone: <input checked="" type="checkbox"/> ?</p> <p>Play Dialing DTMF Tone: <input checked="" type="checkbox"/> ?</p> <p>Auto Answer Tone: <input checked="" type="checkbox"/></p> <p>Ring Back Tone: <input type="text" value="Default"/></p> <p>Open Success Prompting: <input type="text" value="Voice"/></p> <p>Close Door Prompting: <input type="text" value="Voice"/></p> <p>Issuing Failed Prompting: <input type="text" value="Default"/></p> <p>Revoke Failed Prompting: <input type="text" value="Default"/></p>	<p>Enable Call Waiting Tone: <input checked="" type="checkbox"/> ?</p> <p>Play Talking DTMF Tone: <input checked="" type="checkbox"/> ?</p> <p style="border: 2px solid red;">Skip Boot Tone: <input checked="" type="checkbox"/></p> <p>Busy Tone: <input type="text" value="Default"/></p> <p>Open Failed Prompting: <input type="text" value="Voice"/></p> <p>Issuing Success Prompting: <input type="text" value="Default"/></p> <p>Revoke Prompting: <input type="text" value="Default"/></p> <p>Door Sensor Prompting: <input type="text" value="Default"/></p>
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4 Release 2.4.8

Models	Software Version	Comment
i10S/ i10SD/i10SV/ i16S/i16SV/PA3	2.4.8	
PA2S	2.12.4	

4.1 Optimization

12. Improve RTSP video streaming to support TCP transmission
13. Pa2s turns off the startup prompt tone by default
14. Pa2s external LED supports custom settings
15. Add Russian
16. Webpage supports the function of setting alert tone and busy tone options

4.2 Bug Fix

17. Fix the problem that the speed dial key is sometimes pressed invalid (PA3).
18. Fix the problem that the device is connected to a third-party camera and the video screen is not displayed when making a video call with the mobile app.
19. Fix the problem that the device web page cannot switch back to the local camera after switching to the external camera (i16SV).
20. Repair the problem that the device establishes a call, and the auxiliary device holds and cancels the hold for many times; After hanging up and establishing the call again, the auxiliary device will directly display the problem of held.
21. Fix the problem that the device cancels automatic answering and there is no ringing tone when calling.

22. Fix the problem that the device is connected to 4G and Ethernet at the same time, and the account registration is not successful after disconnecting 4G (PA3).
23. Fix the problem that all selectable ringtones will be repeatedly displayed after the ringtone is upgraded on the web page.
24. Fix the problem that the call stream type does not take effect when it is a sub stream.

5 Release 2.4.4

Models	Software Version	Comment
PA3	2.4.4	
PA2S	2.8.11	

5.1 New Features

25. Add the function of converting incoming calls to multicast
26. Add custom LED function
27. Support IPC linkage function
28. Add the function of playing U disk/TF card music through the button
29. Support to turn off the prompt tone of automatic answer
30. Add the ringing timeout time setting to the web page
31. Key event adds volume up/down, loop volume and other types

5.2 Optimization

1. Improve the output trigger without limitation time
2. Improve IP scan support to modify network parameters
3. Improve time plan to play audio to support the number of settings
4. Improve camera settings to support restore to default
5. Improve the server settings under the security settings to add prompt information
6. Improved that after the music is interrupted by the call, the music can be restored after the call is ended
7. Improve Line-in related issues
8. Support closing AEC via HTTP API
9. Support triggering tamper and resetting tamper through active URI

5.3 Bug Fix

1. Fix the problem of poor voice quality when using opus voice codec
2. Fix the problem that the line-in is set to local playback+talking, and the device does not respond after a few minutes of talking
3. Fix the problem that the device is abnormal as the called party in some cases
4. Fix the call timeout time is greater than a certain value, the device does not send a CANCEL when it times out

5.4 New Functions Description

1. Add the function of converting incoming calls to multicast

Function description: Convert the caller's number to multicast and send

2. Add custom LED function

Function description: The web page adds LED settings in various states, and can control the LED lights on, off, and blinking in various states

status light

<p>In Using(Priority goes from high to low)</p> <p>Ring: <input type="text" value="Slowblink"/></p> <p>In Using: <input type="text" value="Slowblink"/></p>	<p>Default Light(Priority goes from high to low)</p> <p>Network Abnormal: <input type="text" value="Fastblink"/></p> <p>SIP Register Fail: <input type="text" value="Fastblink"/></p> <p>SIP Register Success: <input type="text" value="ON"/></p>
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3. Support IPC linkage function

Function description: Support binding third-party cameras

Change the camera mode to external,add the address, user name, password, etc. of the third-party camera. After the setting is successful, you can use the bound camera.

Connection Mode Setting

Native Camera: IP Camera

IP Camera Add

Name:

Username:

Password:

Ip Camera Brand: TOPSEE

IP:

Port:

UserAgent:

URL1:

URL2:

4. Add the function of playing U disk/TF card music through the button

Function description: Add the playback audio type under the DSSKEY setting key event. The setting value is the path of the audio file. Press the set button to play the audio file of the set path. Press again to end; the audio file format supports MP3 and WAV, and the audio file needs to be played. On U disk and TD card

Function Key Settings >>

Key	Type	Name	Value	Value2	Subtype	Line	Media
DSS Key 1	Key Event <input type="button" value="v"/>	<input type="text"/>	<input type="text" value="\usb\music"/>	<input type="text"/>	Audio Play <input type="button" value="v"/>	AUTO <input type="button" value="v"/>	DEFAULT <input type="button" value="v"/>
DSS Key 2	None <input type="button" value="v"/>	<input type="text"/>	<input type="text"/>	<input type="text"/>	None <input type="button" value="v"/>	AUTO <input type="button" value="v"/>	DEFAULT <input type="button" value="v"/>
DSS Key 3	None <input type="button" value="v"/>	<input type="text"/>	<input type="text"/>	<input type="text"/>	None <input type="button" value="v"/>	AUTO <input type="button" value="v"/>	DEFAULT <input type="button" value="v"/>

5. Support to turn off the prompt tone of automatic answer

Function description : Add configuration to enable or disable the prompt tone of automatic answer, the default is enabled

Basic Settings >>

Tone Settings >>

Enable Holding Tone:

Play Dialing DTMF Tone:

Auto Answer Tone:

Enable Call Waiting Tone:

Play Talking DTMF Tone:

Intercom Settings >>

Response Code Settings >>

6. Add the ringing timeout time setting to the web page

Function description: Incoming call automatically hangs up after ringing timeout

7. Key event adds volume up/down, loop volume and other types

Function description : Through the DSSKEY button, you can adjust the volume and other operations