



User Manual

W626W

Software Version: 2.16.30

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3 Safety Instruction

3.1 Safety Instruction

Please read the following safety notices before installing or using this unit. They are crucial for the safe and reliable operation of the device.

- Please use the power adapter specified for the product. In case of special circumstances requiring the use of an adapter from another manufacturer, ensure the voltage and current of the provided adapter comply with the product's specifications. Additionally, it is recommended to use products that have passed safety certifications; otherwise, it may cause fire or electric shock hazards.
- When using this product, do not damage the power cord. Do not forcibly twist, stretch, pull, or tie the power cord; nor should you place heavy objects on it or pinch it between items. Otherwise, the power cord may be damaged, which could lead to fire or electric shock accidents.
- Before using the product, please confirm that the temperature and humidity of its environment meet the product's operating requirements. (When the product is moved from an air-conditioned room to an environment at natural temperature, condensation may form on the product's surface or internal components. Wait for the product to dry naturally before turning on the power.)
- Non-technical service personnel must not disassemble or repair the product by themselves. Improper repair may cause accidents such as electric shock or fire, and the warranty service of your product will also become invalid.
- Do not insert metal foreign objects such as pins or iron wires into the vents or gaps of the product. Otherwise, it may cause injury accidents like electric shock due to electric current passing through the metal foreign objects. If any foreign objects or similar metal items fall into the product, stop using it immediately.
- Do not discard or store the plastic bags used for packaging in places accessible to young children. This is to prevent young children from putting the plastic bags over their heads, which may block their noses and mouths and cause suffocation.
- Please use this product correctly in accordance with the instructions provided in the product manual. Prolonged improper operation may cause damage to the product and pose potential safety hazards.

3.2 Battery Warning

To prevent the battery pack from leaking, overheating and catching fire. Please observe the following precautions:

- Soft aluminum packaging foil is easily damaged by sharp edged parts such as nickel sheets, pins and needles.
- Do not hit the battery pack with any sharp parts.
- Do not immerse the battery pack in water or seawater.
- Do not use and place the battery pack near heat sources such as fire or heaters.

- Do not reverse the positive and negative poles.
- Do not connect the battery pack to the power outlet.
- Do not throw the battery pack into fire or heat.
- Do not connect the positive and negative poles directly to metal objects such as wires, which may cause a short circuit of the battery pack.
- Do not transport and store batteries together with metal items such as necklaces and hairpins.
- Do not knock or throw the battery pack.
- Do not directly solder the battery pack or battery, and do not pierce the battery with nails or other sharp objects.
- Please use the charger configured at the time of purchase when charging. [Product Overview](#).

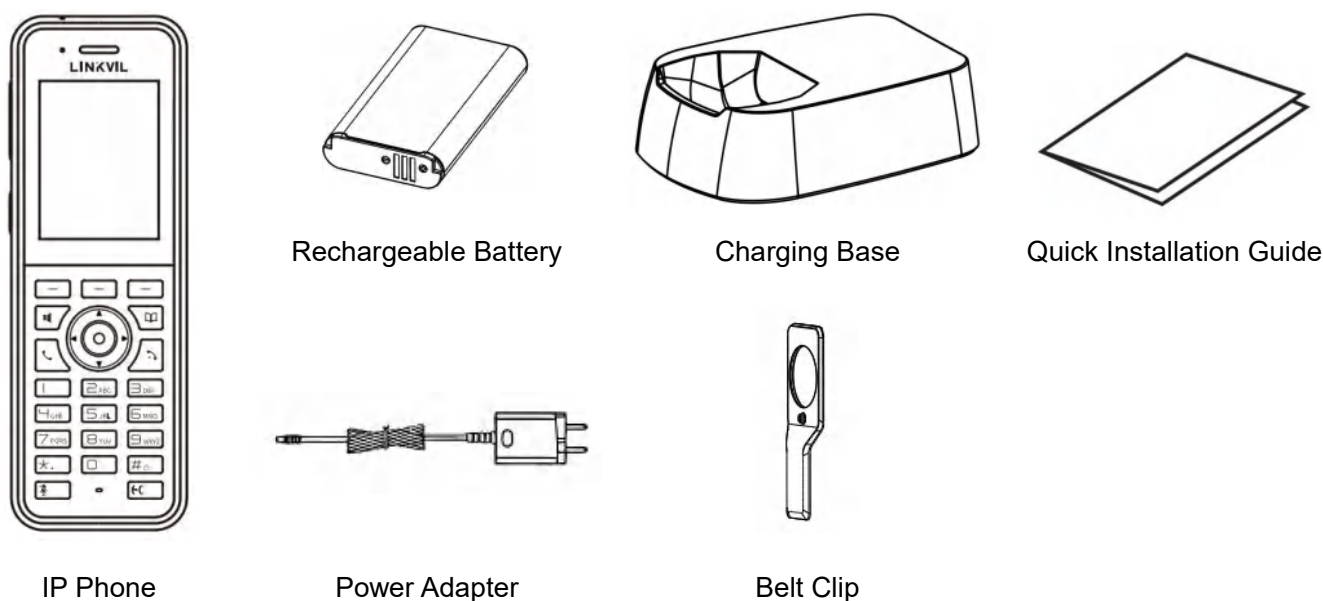
4 Overview

4.1 Overview

W626W is an enterprise-class portable Wi-Fi color IP phone designed for professional mobile communications. The product has IP67 protection level, antibacterial and chemical corrosion resistance, waterproof and dust-proof, and can withstand a drop impact of 1.8 meters. It supports up to 4 SIP accounts, 6 local meetings, and is equipped with a 2.4-inch color display. It integrates Bluetooth 5.0 and dual-band 2.4G / 5G Wi-Fi 6 and supports seamless roaming. It can also ensure stable transmission of voice and data in complex wireless environment. Equipped with high-definition audio and intelligent noise reduction technology, clear and reliable voice communication is realized. With flexible charging methods, 3100 mAh rechargeable battery can meet the long-term endurance requirements of all-weather use, and are widely used in medical, manufacturing, warehousing, transportation, education, small and medium-sized enterprises, hotels and outdoor operations.

To help interested users gain a better understanding of the product details, this user manual can serve as a reference guide for device usage. This document may not apply to the latest software version. If you have any questions, you can use the help prompt interface built into the device phone, or download and update your user manual from <https://fanvil.com/>.

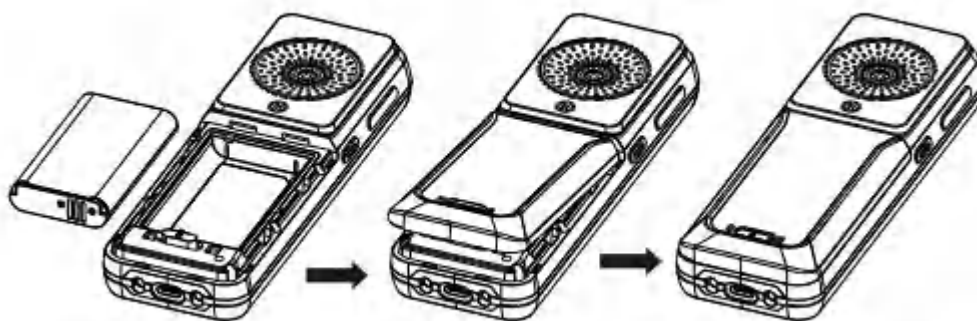
4.2 Packing Contents



Picture 1 - packing contents

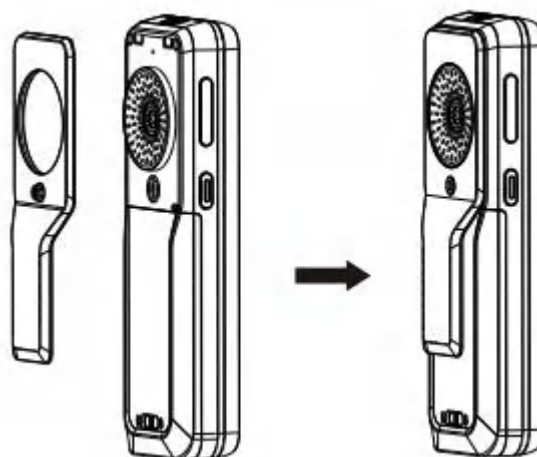
5 Desktop Installation

5.1 Install the battery



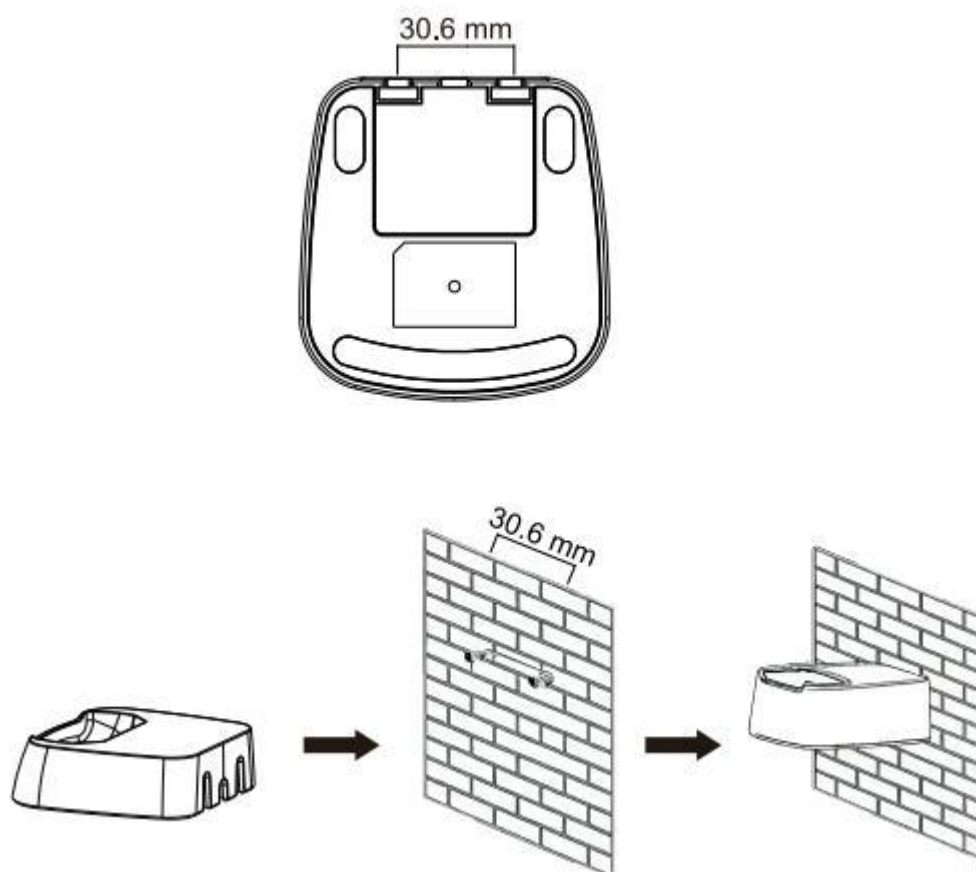
Picture 2 - Install the battery

5.2 Install the belt clip



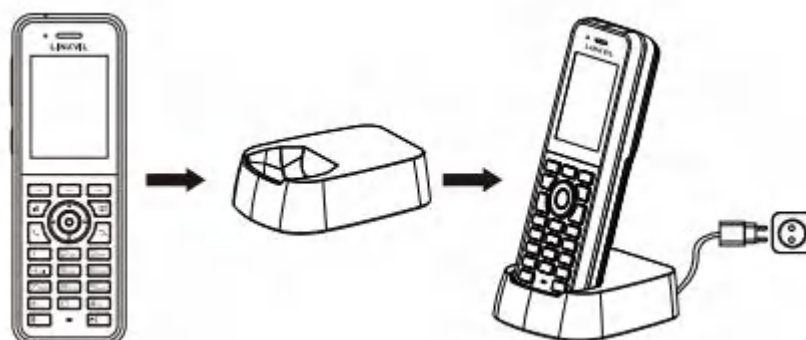
Picture 3 - Install the belt clip

5.3 Install the charging base on the wall (optional)

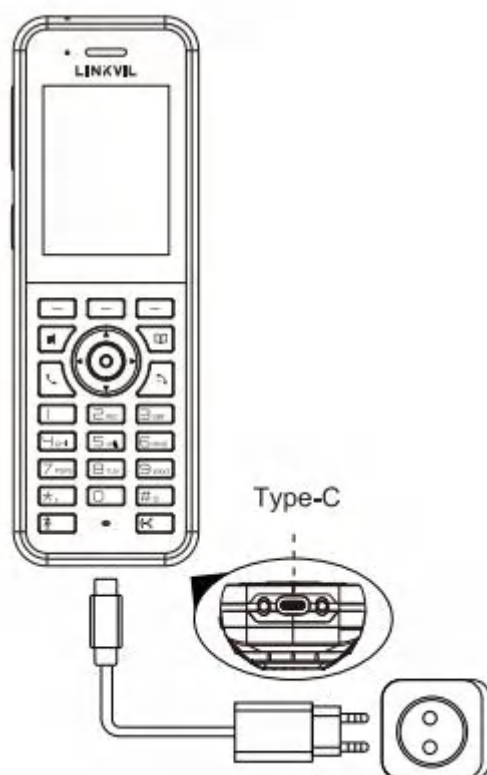


Charging methods:

1. Put the Wi-Fi phone on the charging base and turn it on.



2. Charging Wi-Fi via the Type-C interface at the bottom of the device

**Note:**

1. Use Fanvil-supplied power adapter (5V/2A). A third-party power adapter may damage the handset.
2. Check the charging status on the top-right corner of the LCD screen.

6 Appendix Table

6.1 Appendix I - Icon

Table 1 - Keypad Icons

Icon	Description
	Programmable keys, the default is PTT key, which can be set to function key/speed Dial key, etc.
	Volume key, in standby or ringing mode: increase or decrease ringtone volume In call mode: increase or decrease the headset, handset, and hands-free volume
	Soft keys
	Up, down, left, right navigation keys, and OK key By default, the functions of each key during standby: Up key: Call history DSS Key Down key: status DSS Key Left key: Enter the DSS Key Page1 settings page Right key: Enter the DSS Key Page2 settings page OK key: Long press to restore factory settings, press the [OK] key to enter the menu when the device is in standby mode..
	Return to the homepage; Hang up the call; Entering sleep mode; Exit sleep mode Long press to power on/off
	Dialing; Switching call channels
	Phone book keyboard
	Hands-free key, enable/disable the hands-free channel
	Mute key, turn on/off mute
	Call transfer key, during a call can transfer the call to another number

Table 2 - Status Prompt and Notification Icons

Screen Icon	Description
	In hands-free mode
	In headset mode
	In handset mode
	Mute activated
	Silent mode
	Call is on hold
	Auto-answering activated
	Call forward activated
	Do not disturb activated (Red)
	SIP hotspot activated
	VPN activated
	Bluetooth device paired connection
	Voice quality level of call
	Keypad locked
	Forward call(s)
	Missed call(s)
	Received call(s)

	Dialed call(s)
	Wireless network connected
	Wireless network disconnected
	Wireless network failure

6.2 Appendix II - Keyboard character query table

Table 3 - Look-up Table of Characters

Mode Icon	Text Mode	Key key	Characters Of Each Press
	Numeric		1
			2
			3
			4
			5
			6
			7
			8
			9
			0
			*.:/@[],+_-='?\"';()<>{}#
			#
	Lower Case Alphabets		@:;()<>[]{}
			a b c
			d e f
			g h i
			j k l
			m n o
			p q r s
			t u v
			w x y z
			(space)
			.,*/+-.:_'?\" #!&\$%£¥☒~¡¿\$

		#	#^!&\$%£¥☒~¡ ¢ §
ABC	Upper Case Alphabets	1	@:;()<>[]{}
		2	A B C
		3	D E F
		4	G H I
		5	J K L
		6	M N O
		7	P Q R S
		8	T U V
		9	W Z Y X
		0	(space)
		*	.,*/+:-_='?\" #^!&\$%£¥☒~¡ ¢ §
		#	#^!&\$%£¥☒~¡ ¢ §
2aB	Mixed type input	1	1
		2	2 a b c A B C
		3	3 d e f D E F
		4	4 g h I G H I
		5	5 j k l J K L
		6	6 m n o M N O
		7	7 p q r s P Q R S
		8	8 t u v T U V
		9	9 w z y x W Z Y X
		0	0
		*	@:;()<>[]{}.,*/+:-_='?\" #^!&\$%£¥☒~¡ ¢ §
		#	#^!&\$%£¥☒~¡ ¢ §
Abc	首字母大写字母键盘	1	@:;()<>[]{}
		2	单词首字母: A B C 非首字母: a b c
		3	单词首字母: D E F 非首字母: d e f
		4	单词首字母: G H I 非首字母: g h i
		5	单词首字母: J K L 非首字母: j k l
		6	单词首字母: M N O 非首字母: m n o

		7	单词首字母: P Q R S 非首字母: p q r s
		8	单词首字母: T U V 非首字母: t u v
		9	单词首字母: W X Y Z 非首字母: w x y z
		0	(space)
		*	.,*/+:-!_=' ?\" #^!&\$%£¥☉~¡ ¢ §
		#	#^!&\$%£¥☉~¡ ¢ §

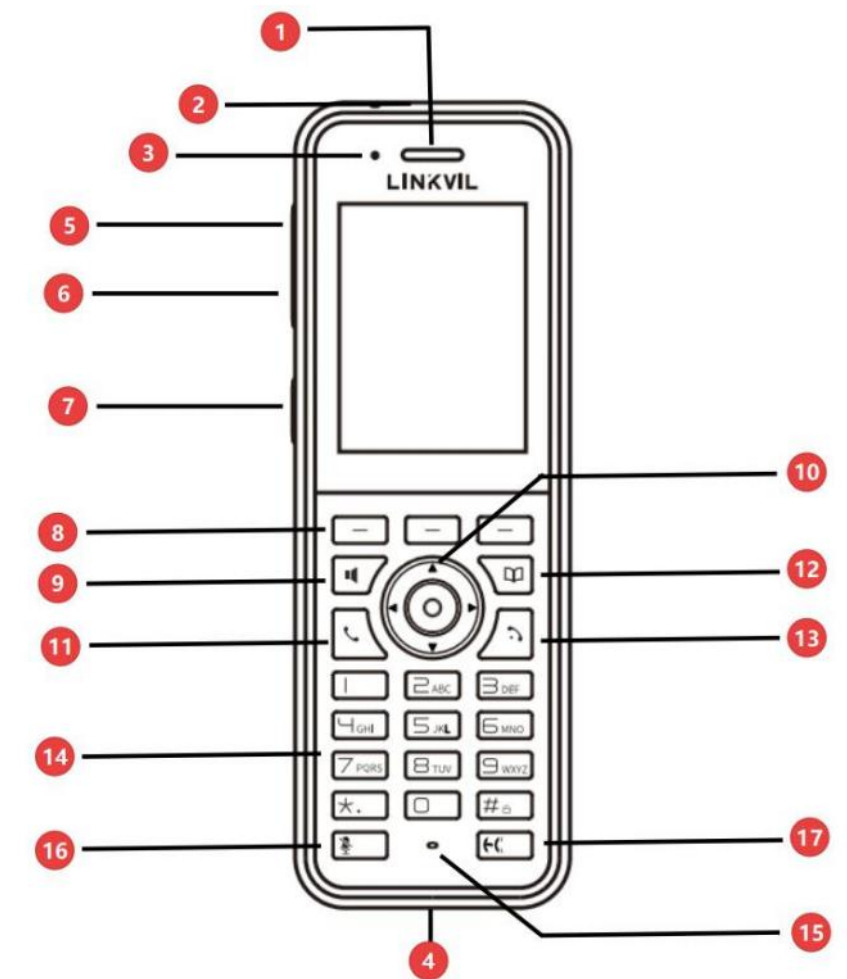
6.3 Appendix III - LED Definition

Table 4 - Status Indicator LED State

Type	LED Light	State
Status Indicator	Orange On	Charging
	Green On	Fully Charged
	Red and Orange Blinking alternately	Low Battery/Abnormal charging
	Orange Breathing Blinking	Missed calls or unread messages
	Orange Blinking Quickly	Ringing
	Red Blinking Quickly	POST Mode Prompt

7 Introduction to the User

7.1 Instruction of Keypad



Picture 4 - W626W key layout

Table 5 - Instruction of Keypad of W626W

Index	Key Name	Description
①	Earpiece	Play sound when talking in handset mode.
②	Alarm Trigger Key	Triggers an alarm when pressed manually.
③	Status indicator	Indicates the status of the device during calls and charging.
④	Charging interface	Users can charge the handset here via the charging base or power adapter, and it also serves as a Type-C wired headphone jack.
⑤	Volume Key +	In standby mode, on the ringing and ringing configuration interface, press this key to increase the volume of the ringtone; On the call or volume adjustment interface, press this key to

		increase the volume.
⑥	Volume Key -	In standby mode, on the ringing and ringing configuration interface, press this key to decrease the volume of the ringtone; Press this key to decrease the volume on the call or tone/brightness adjustment interface.
⑦	PTT key	Before configuration, short/long press to enter the function key setting interface and configure the required functions; After configuration, press to implement the corresponding function.
⑧	Soft key	These two keys provide corresponding menu functions displayed on the screen. This key is designated as DSS Key on the handset's standby interface.
⑨	Hands-free	Users can press this key to turn on the audio channel of the hands-free speaker.
⑩	Navigation key	On some settings and text editing pages, users can press the Left/Right/Up/Down navigation keys to change options or move the cursor Left/Right/Up/Down in the screen list. Users can configure different functions for navigation keys on the Manager website.
⑪	Off-hook key	Dialing; Switch call channels.
⑫	Phonebook	By pressing the "PhoneBook" key, users can enter the device book interface and select a contact person to make a call.
⑬	On-hook key/Power	Return to the homepage; Hang up the call; Enter sleep mode; Exit sleep mode; Long press to enable/restart.
⑭	Keypad	These 12 standard keys provide standard device key functions. At the same time, long pressing certain keys can trigger the provision of special functions. Lone press [#] key to lock the device (enable keyboard lock configuration).
⑮	Microphone	Receive audio during calls.
⑯	Mute key	Press this key on the standby interface to mute the incoming call ringtone; Press this key on the call interface to mute the call.
⑰	Call transfer key	Press this key during a call can transfer the call to another number.

7.2 Using Handset / Hands-free Speaker / Headset

■ Using Handset

For using the handset: Enter the number via the **[Dial key **], then press the **[Dial key **] / **[#]** key / **[soft key-dial]** to make the call; alternatively, press the **[Dial key **] after entering the number to initiate the call. When the speakerphone is on, press the **[Dial key **] to switch the device's audio channel.

■ Using Hands-free Speaker

For using the hands-free speaker: Either press the **[Hands-free key **] first then make the call, or enter the number first followed by pressing the **[Hands-free key **]. When the audio channel of the handset is active, press the **[Hands-free key **] to switch the device's audio channel.

■ Using Headset

In the factory settings, after inserting headset, the headset channel is turned on by default. When the headset is plugged in, the user can press the **[Dial key]** to enter the number, and then press the **[Dial key **] / **[#]** key; Or enter the number first and then press the **[Dial key **]. The number will be dialed out and a call will be established through the headset channel. When the speaker is turned on, the user can switch the audio channel of the phone by pressing the **[Dial key **].

7.3 Power on/off operation

Shutdown operation

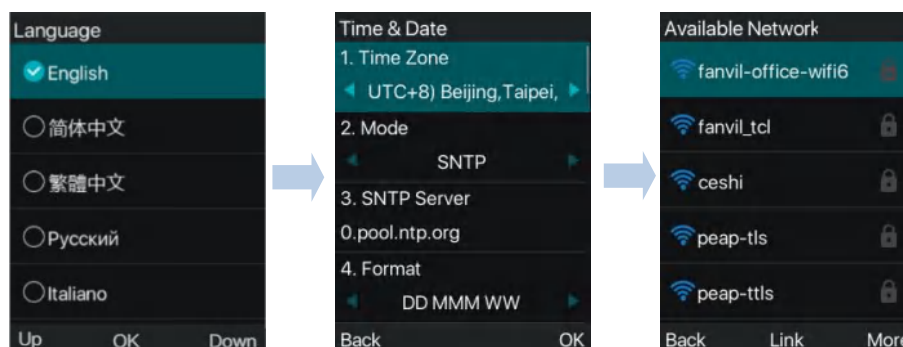
- Shutdown: long press the **[Red On-hook key **] to pop up the shutdown window, choose whether to shutdown.
- Automatic shutdown when low battery: When the battery is about to run out, it will prompt "low battery". The phone automatically shuts down when the battery runs out.

Reboot

- Press **[OK]** key in standby mode to enter menu >> **[Settings]** >> **[Reboot System]**, and select **[OK]** to reboot.
- long press the **[Red On-hook key **] until reboot.

Boot operation

- Long press the **[Red On-hook key **] until the device turns on, or turn it on by connecting a charger.
- There is a setup wizard when the device is powered on under factory settings:
 - Select the language, time & date in sequence, then connect to the network.



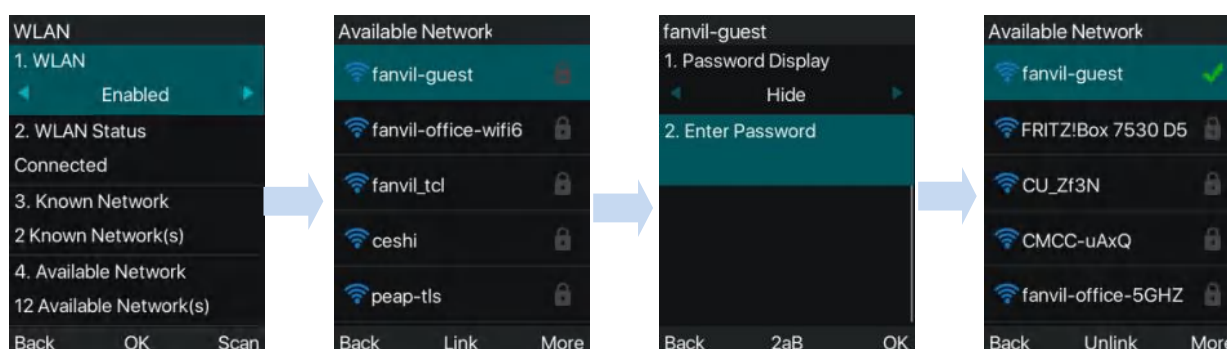
Picture 5 - W626W setup wizard

7.4 Connection Wi-Fi

Users can connect to Wi-Fi through the device or the web.

● Device side

1. Press **[OK]** key in standby mode to enter menu >> **[WLAN]**, and enable the wireless network.
2. Select an **[Available Network]** and click **[Enter]**.
3. Select the corresponding Wi-Fi network from the Wi-Fi list, click **[Link]**, and enter the password.
4. After clicking **[OK]**, the Wi-Fi connection is successful, and the corresponding Wi-Fi display will show **✓**.



Picture 6 - Wi-Fi connection on the device side

● web side

Log in to the device web page, and go to **[Network]** >> **[Wi-Fi Settings]**. Select to enable Wi-Fi, and configure the corresponding SSID, username and password.

Wi-Fi Settings

Wi-Fi Enable: ☒ Band Mode:

Roaming Delta: (3~30)dB Roaming Trigger: (-70~-50)dBm

Roaming Scan Period: (1~3600)s Full Roam Scan Period: (1~3600)s

Wi-Fi Country Code:

Wi-Fi Info Add

Wi-Fi Name:

SSID:

Secure Mode:

Wi-Fi Info List

☐	SSID	Secure Mode	Encryption Type
☐	WiFi-device-ssid	WPA/WPA2-PSK	None
☐	fanvil-office-5GHZ	802.1x PEAP-MSCHAPv2	None

Picture 7 - Web connection Wi-Fi

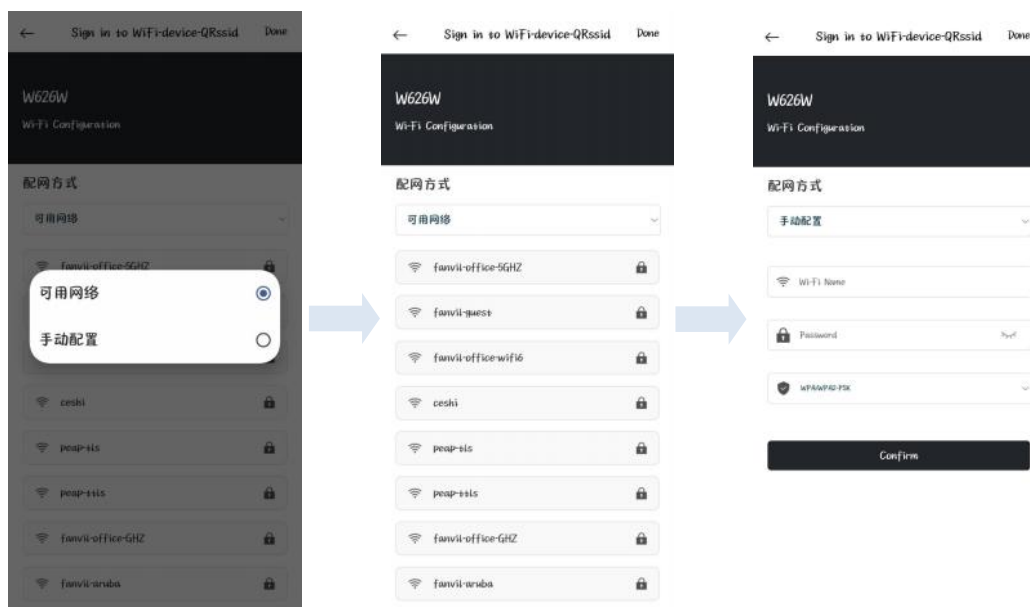
● Wi-Fi Quick Connection

- On the handset's standby screen, press the **[OK]** key to enter the menu >> **[WLAN]** >> **[Wi-Fi Quick Connection]**, click to **[Enter]**, and a QR code will be displayed.



Picture 8 - Wi-Fi Quick Connection

- Scan the Wi-Fi QR code with your mobile phone, and you can select **[Available Networks]** or **[Manually configure]** in the Network Setup.
 - ◆ After selecting **[Available Networks]**, the mobile phone will display the list of scanned WiFi networks. Users can select the network to connect to, enter the password, choose the network mode, and then connect to the network.



Picture 9 - Mobile phone scanning the Wi-Fi QR code - Available Networks

- ◆ After selecting **[Manually configure]**, enter the Wi-Fi SSID, password, and security mode to connect to the network.



Picture 10 - Mobile phone scanning the Wi-Fi QR code - Manually configure

7.5 Connect to Bluetooth

W626W supports Bluetooth external line.

The method of connecting W626W to a Bluetooth device is as follows:

1. Press **[OK]** key in standby mode to enter menu >> **[Basic]** >> **[Bluetooth]** to enable Bluetooth settings.
2. Select the **[Paired Device]** and press the **[OK]** key to scan for available Bluetooth around.
3. Select the corresponding Bluetooth device in the Bluetooth list, click **[Link]**, the screen will prompt the

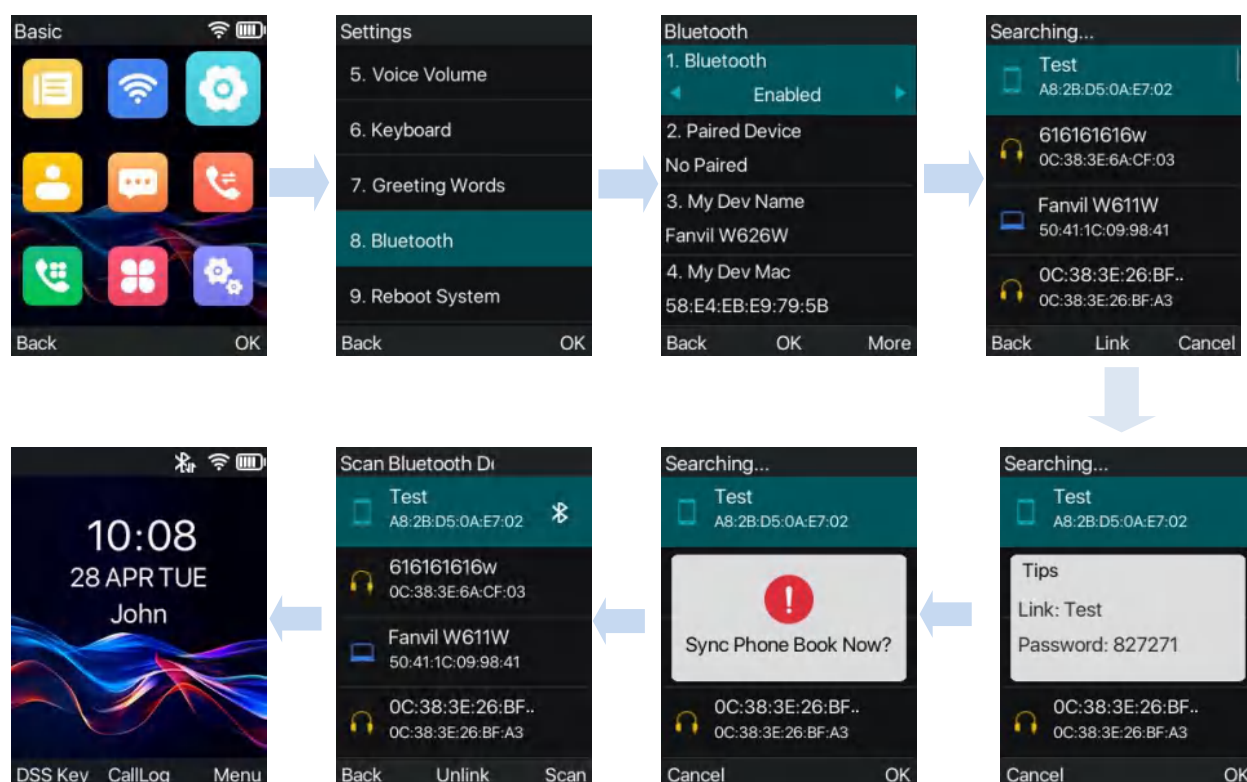
pairing key.

4. After the pairing confirmation is clicked on both the corresponding Bluetooth device and W626W, the Bluetooth connection is successful.

5. If the Bluetooth device is a mobile phone, W626W will prompt "Sync Phone Book Now?" After confirming, the phone contacts will be synchronized to the W626W contact directory, which can be viewed in **[Menu] >> [Contact] >> [Mobile Contacts]**.

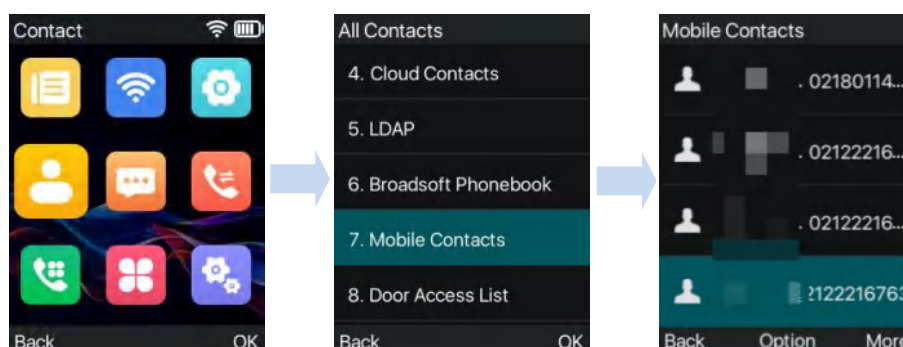
6. After the Bluetooth is connected, the Bluetooth icon  will be displayed after the Bluetooth device in the Bluetooth list. At the same time, the status bar of the standby interface will also display the Bluetooth icon.

Connect to Bluetooth



Picture 11 - Connect to Bluetooth

View Synced Mobile Contacts



Picture 12 - Sync mobile phone contacts

7.6 Sleep State

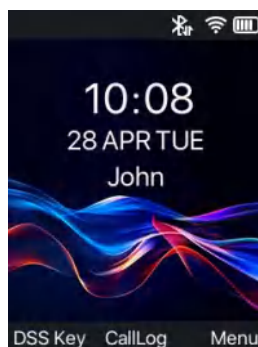
Enter sleep state

- When the W626W is in no operation state, it enters the sleep state by default for 30 s.
- Standby press the **[Red Hang-up key **] to enter the sleep state.

Exit sleep state

- The phone automatically exits the sleep state when there is an incoming call.
- Press the phone key to exit the sleep state.

7.7 Standby Interface



Picture 13 - Screen Layout / Default Main Interface

The image above shows the default idle screen interface, which is the state of the user interface most of the time.

The status bar on the main screen displays the device status, information and data that can be edited (such as voice messages, missed calls, auto answer, do not disturb, locked status, network connection status, etc.).

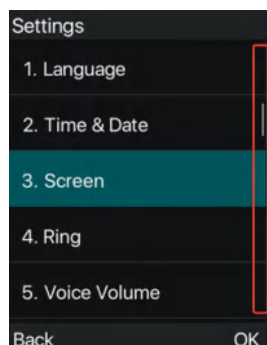
The middle area shows the time and the registration status of the account.

The area below is the function menu keys, which is also the first layer of the function menu keys, through which the user can operate the phone.

The user can restore the phone to the default idle screen interface by pressing the **[On-hook key **].

Icon description is described in [6.1 Appendix I](#).

On some screens, the screen does not fully display due to too many items or too long text. They will be grouped together into a list with scroll icons. If the user sees a scroll icon, they can use the **[Up/Down navigation keys]** to scroll through the list.



Picture 14 - Scroll Icon

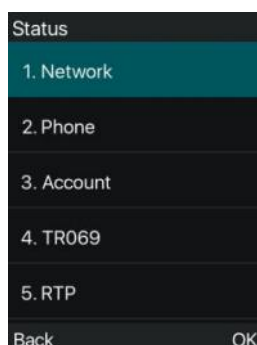
7.8 Phone Status

The phone status includes the following information about the phone:

- Network Status:
 - IPv4 or IPv6 status
 - IP Address
 - Network mode
 - Wi-Fi Name
 - BSSID
 - Signal Strength
- The Phone Device Information:
 - Wi-Fi MAC address
 - Bluetooth MAC address
 - Phone Mode
 - Hardware Version number
 - Software Version number
 - Phone Storage (RAM and ROM)
 - System Running Time
 - UBoot
- SIP Account Information:
 - SIP Account
 - SIP Account Status (register / uncommitted / trying / time out)
- TR069 Connect Status (Displays only in the phone interface state)

The user can view the phone status through the phone interface and the web interface.

- Phone interface: When the phone is in standby mode, press **[OK]** key in standby mode to enter menu >> **[Status]** and select the option to view the corresponding information, as shown in the figure:



Picture 15 - The Phone status

- WEB interface: Refer to [7.9 Web management](#) to log in the phone page, enter the **[System]** >> **[Information]** page, and check the phone status, as shown in the figure:

System Information ?	
Model:	W626W
Hardware:	V1.1
Software:	0.0.13
Uboot:	V1.0
Uptime:	29 : 47 : 47
MEMInfo:	ROM: 12.1/128(M) RAM: 36.1/128(M)
System time:	16:29 26 MAY TUE (SNTP)
Network ?	
WAN	
Network mode:	DHCP
Wi-Fi MAC:	0c:38:3e:90:4a:7f
Bluetooth MAC:	58:e4:eb:e9:79:5b
IPv4	
Wi-Fi IP:	172.16.2.201
Subnet mask:	255.255.254.0
Default gateway:	172.16.2.1
VO status ?	

Picture 16 - WEB phone status

7.9 Web Management

Phone can be configured and managed on the web page of the phone. The user needs to enter the IP address of the phone in the browser and open the web page of the phone firstly. The user can check the IP address of the phone by pressing **[OK]** key in standby mode to enter menu >> **[Status]** >> **[Network]** >> **[Wi-fi IP]**.

Open the browser, enter the phone IP, log in to the phone web page, the first thing to see is the phone login page.



Picture 17 - Landing page

Users must correctly enter the user name and password to log in to the web page. **The default username and password are both "admin"**. For the specific details of the operation page, please refer to [11Web Configurations](#).

7.10 SIP Configurations

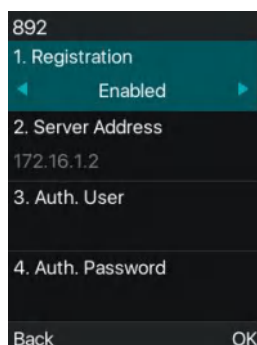
At least one line must be configured properly to be able to provide telephone service. The line configuration is like a virtualized SIM card on a mobile phone which stores the service provider and the account information used for registration and authentication. When the device is applied with the configuration, it will register the device to the service provider with the server's address and user's authentication as stored in the configurations.

The user can conduct line configuration on the interface of the phone or the web page, and input the corresponding information at the registered address, registered user name, registered password, SIP user and registered port respectively, which are provided by the SIP server administrator.

- **Phone interface:** To manually configure a line, the user can press the line key for a long time, or press [OK] key in standby mode to enter menu >> [Advanced] >> [Accounts] >> [Line n] >> press [OK] >> [Basic] to configuration, click [OK] to save the configuration.

NOTICE! User must enter correct PIN code to be able to System to edit line configuration. (The default PIN is 123)

The parameters and screens are listed in below pictures,.



Picture 18 - Phone line SIP address and account information

- **WEB interface:** After logging into the phone page, enter [Line] >> [SIP] and select SIP for configuration, click apply to complete registration after configuration, as shown below:

Line 1008@SIP2

Register Settings >>

Line Status:	Registered	Activate:	☒ ?
Username:	1008 ?	Authentication User:	1008 ?
Display name:	1008 ?	Authentication Password:	***** ?
Realm:	?	Server Name:	?

SIP Server 1:		SIP Server 2:	
Server Address:	172.16.1.44 ?	Server Address:	?
Server Port:	5060 ?	Server Port:	5060 ?
Transport Protocol:	UDP ?	Transport Protocol:	UDP ?
Registration Expiration:	3600 second(s) ?	Registration Expiration:	3600 second(s) ?
Proxy Server Address:	?	Backup Proxy Server Address:	?
Proxy Server Port:	5060 ?	Backup Proxy Server Port:	5060 ?
Proxy User:	?		
Proxy Password:	?		

Picture 19 - Web SIP registration

8 Basic Function

8.1 Making Phone Calls

■ Default Line

The phone provides multiple SIP line services. If both lines are configured, user can make or receive phone calls on either line. If default line is configured by user, there will be a default line to be used for making outgoing call which is indicated on the intermediate area. To enable or disable default line, user can press **[OK]** to enter **Menu >> [Features] >> [General] >> [Default Line]** or configure from Web Interface (**[Web] >> [Phone settings] >> [Features] >> [Basic Settings]**).



Picture 20 - Default line

■ Dialing Methods

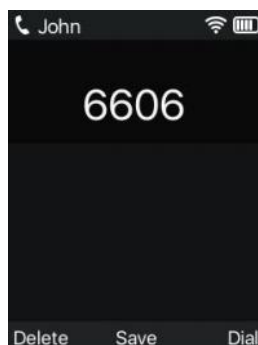
User can dial a number by,

- Entering the number directly
- Selecting a phone number from local contacts (Refer to [10.2.1 Local Contact](#))
- Selecting a phone number from cloud contacts (Refer to [10.2.3 Cloud Contact](#))
- Selecting a phone number from call logs (Refer to [10.3 Call Log](#))
- Redialing the last dialed number

■ Dialing Number then Opening Audio

To make a phone call, user can firstly dial a number by one of the above methods. When the dialed number is completed, user can press **[Dial]** key on the soft-menu, or press **[Hand-free key 

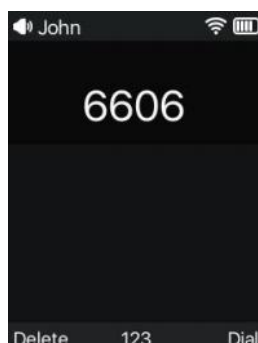
36 / 139**



Picture 21 - Enable voice channel dialing

■ Opening Audio then Dialing the Number

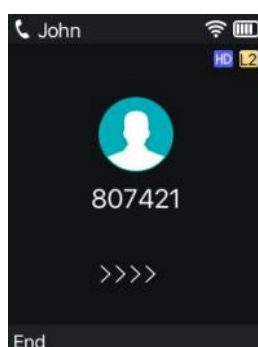
Another alternative is the traditional way to firstly open the audio channel by lifting the handset, and then press the **[Dial key 📞]** or **[Hands-free key 📞]**, dial the number with one of the above methods. When completing the number dial, user can press **[soft key-Dial]** key / **[OK]** key / **[Dial key 📞]** to call out, or wait seconds to call out.



Picture 22 - Open the voice channel and dial the number

■ Cancel Call

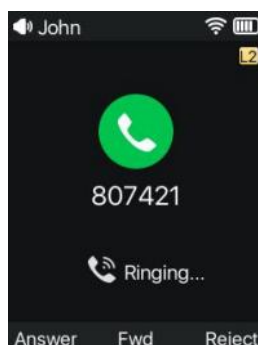
When making a call, the user can press the **[soft key-End]** or **[Hang-up key 📞]** to close the voice channel.



Picture 23 - Call number

8.2 Answering Calls

When the idle device receives an incoming call, the below incoming call appears on the screen.

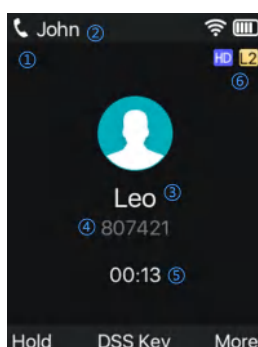


Picture 24 - Answering calls

User can answer the call by turning on the handset/headset/hands-free channel by pressing the **[Dial key ]** / **[Hands-free key .** To reject an incoming call, the user should press the **[Reject]** / **[Hang up key .**

8.2.1 Talking

When a call is connected successfully, user will see a talking mode screen as the following figure.



Picture 25 - Talking interface

Table 6 - Talking mode

Number	Name	Description
①	Voice channel	The icon shows the voice channel mode being used.
②	Default line	The line currently in use by the phone.
③	Call-to-end name	The name of the person on the other end of the call. When the phone initiates an outgoing call, this position displays the remote number instead of the contact name.
④	Call-to-end number	The number of the person on the other end of the call. When the phone initiates an outgoing call, this area remains blank.

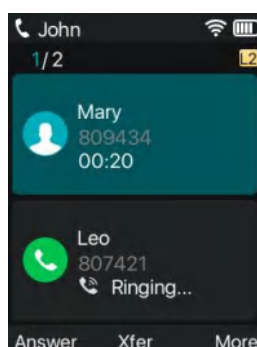
⑤	Call duration	The duration of a call after it has been established.
⑥	Speech quality	Displays the current voice quality of the call.

8.2.2 Make / Receive Second Call

The phone supports multiple calls. When there is already a call established, user can still answer another incoming call on either lines or make a second call on either lines.

■ Second Incoming Call

When there is another incoming call during talking a phone call, this call will be waiting for user to answer. User will see the call message in the middle of current screen. The device will not be ringing but playing call waiting tone in the audio channel of the current call and the LED will be flashing in orange. User can accept or reject the call as same as normal incoming call. When the waiting call is answered, the first call will be held on automatically.



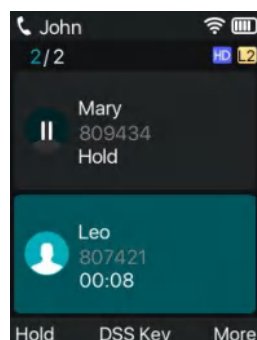
Picture 26 - The second call interface

■ Second Outgoing Call

To make a second call, User can press **[More]** >> **[New Call]** / **[Transfer]** / **[Conf]** or the physical key-**[Call transfer]** to dial the second call. Then dial the number the same way as making a phone call. Another alternative for making second call is to press DSS Keys or dial out from the configured Keys (BLF/Speed Dial). When the user is making a second call with the above methods, the first call could be held on manually or will be held on automatically at second dial.

■ Switching between Two Calls

When there are two calls established, user will see a dual calls screen as the following figure.



Picture 27 - Two way calling

User can press **[Up/Down navigator keys]** to switch screen page, and switch call focus by pressing **[More] >> [Resume]**.

■ Ending One Call

User may hang up the current talking call by closing the audio channel or press **[End]** key. The device will return to single call mode in holding state. Users can also resume the current call by pressing the **[More] >> [Resume]** key.

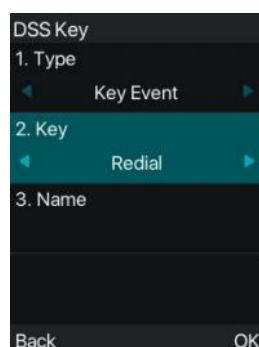
8.3 End of the Call

After the user finishes the call, the user can press the **[Hang up key 📞]** or **[soft key-End]** to close the voice channel and end the call.

Note! When the phone is in the reserved state, the user could press the [soft key-Resume] to return to the call state, or press [More] >> [End] or the [Hang up key 📞] to hang up.

8.4 Redial

W626W has no physical redial key, users can set the side function key as redial key when using. Setting method: In standby, press **[OK]** key in standby mode to enter menu >> **[Basic] >> [Keyboard] >> [DSS Key Settings]**, select the DSS Key to be set, select the type of DSS Key for the type, and select redial for the name of the DSS Key. You can set the name of the callback key in the name input box, and press the **[OK]** key to save.



Picture 28 - Set redial DSS key

- Redial the last outgoing number:
When the phone is in standby mode, press the redial key and the phone will call out the last outgoing number.
- Call out any number with the redial key:
Enter the number, press the redial key, and the phone will call out the number on the dial.

8.5 Dial-up Query

Dial-up query function is enabled on the handset by default. Press the Off-hook key or **[Hands-free**

key] to enter the dialing interface. When one or more numbers are entered, the interface will automatically match phone numbers from call logs and contacts. Use the **[Up/Down navigation keys]** to select a number, then press the **[Dial]** / **[Dial key**] / **[Hands-free key**] to make a call.

8.6 Auto-Answering

The user can enable the auto-answer function on the phone. When an incoming call arrives with auto-answer enabled, the phone will answer the call automatically. This function can be enabled per line, and may be turned on or off via the phone or web management page.

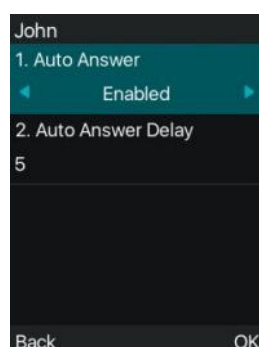
- **Phone interface:**

Press **[OK]** key in standby mode to enter menu >> **[Features]** >> **[Auto Answer]**;

Select the line, use the **[Left/Right navigation key]** to turn on/off the auto answer option, and set the auto answer time. The default auto answer time is 5 seconds.

After completion, press **[OK]** key to save;

The icon in the upper right corner of the screen indicates that auto answer is enabled.



Picture 29 - Line enables auto-answering



Picture 30 - The line has enabled auto-answering

- **WEB interface:**

Log in the phone page, enter **[Line]** >> **[SIP]**, select a SIP line >> **[Basic settings]**, check “Enable Auto Answering” and click “apply” after setting the auto answering time.

Line: John@SIP2

Register Settings >>

Basic Settings >>

Enable Auto Answering: ☒ ?

Auto Answering Delay: 5 (0~120)second(s) ?

Call Forward Unconditional: ☐ ?

Call Forward on Busy: ☐ ?

Call Forward on No Answer: ☐ ?

Call Forward Number for Unconditional: ?

Call Forward Number for Busy: ?

Call Forward Number for No Answer: ?

Picture 31 - Web page to start auto-answering

8.7 Callback

The user can dial back the number of the last call. If there is no call history, press the **[Callback]** key and the phone will say "can't process".

- Set the callback key through the phone interface:
Under standby, press **[OK]** key to enter menu >> **[Basic]** >> **[Keyboard]** >> **[DSS Key Settings]** or the physical key **[DSS Key]** .Choose **[Call Back]** function, input the callback key name, press **[OK]** key to save.

DSS Key

1. Type
Key Event

2. Key
Call Back

3. Name

Back OK

Picture 32 - Set the callback key on the phone

- Set the callback key through the web interface:
Log in the phone page, enter the **[Function Key]** >> **[Function Key]**, select the function Key, set the type as the function Key, and set the subtype as the callback, as shown in the figure:

Function Key Settings

Dsskey Transfer Mode: Make a New Call

Dsskey Edit: Short Press/Long P

Apply

Page1 Page2

Delete Add New Page

Key	Type	Name	Value	Subtype	Line	Media	PickUp Number
DSS Key 1	Key Event			Call Back	AUTO	DEFAULT	
DSS Key 2	None			None	AUTO	DEFAULT	
DSS Key 3	None			None	AUTO	DEFAULT	
DSS Key 4	None			None	AUTO	DEFAULT	

Picture 33 - Set the callback key on the web page

8.8 Mute

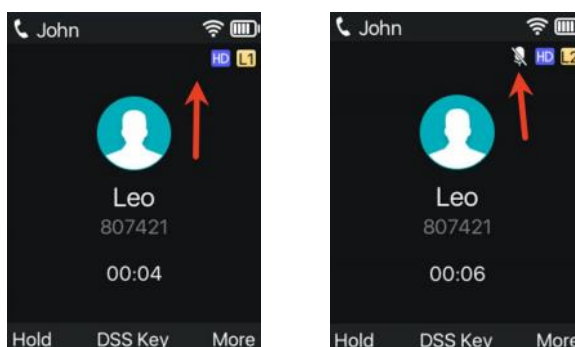
The user can turn on mute mode during a call to turn off the microphone so that the local voice will not be heard. Normally, mute mode will be automatically turned off at the end of a call. The user can also turn on mute when the phone is in standby and mute the ringtone automatically when there is an incoming call.

Mute mode can be turned on in all call modes (handset, headsets or hands-free).

8.8.1 Mute the Call

- Press **[More]** >> **[Mute]** or the physical key **[Mute key]** during a call.

Mute icon  is displayed in the call interface, as shown in the figure:



Picture 34 - Mute the call

- Cancel mute: Press **[More]** >> **[Mute]** or the physical key **[Mute key]** again. By doing so, the local sound will be heard again and the mute icon will no longer be displayed before mute mode is enabled again.

8.8.2 Ringing Mute

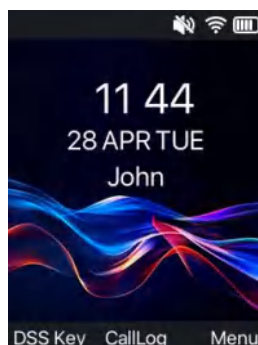
Users can set shortcut key function key as mute key: Under standby, press **[OK]** key to enter menu >> **[Basic]** >> **[Keyboard]** >> **[DSS Key Settings]** or the physical key **[DSS Key]**. Choose **[Mute]** function, input the callback key name, press **[OK]** key to save.

Mute can also be turned on/off with the **[volume up and down keys]** or the physical key **[Mute key]**.

Turn on ringing mute

- Press the **[DSS Key-Mute]** or the physical key **[Mute key]** when the phone is in standby mode.
- Press the **[volume down key]** for several times to mute when the phone is in standby

When mute mode is enabled, the top right corner of the phone will show the bell mute icon , when there is an incoming call, the phone will display the incoming call interface but will not ring.



Picture 35 - Ringing mute

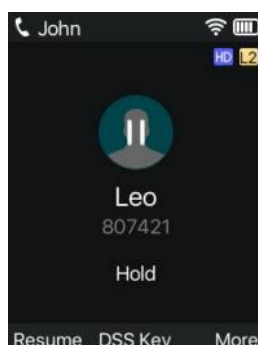
Cancel ring tone mute:

- Press the **[Mute]** function key or the physical key **[Mute key]** when the phone is in standby mode.
- Press the **[volume up key]** to mute when the phone is in standby.

The mute icon  will no longer be displayed in the upper right corner after canceling.

8.9 Call Hold/Resume

The user can press **[More]** >> **[Hold]** to maintain the current call, and this key will switch to **[More]** >> **[Resume]**, the user can press the **[More]** >> **[Resume]** to restore the call.



Picture 36 - Call hold interface

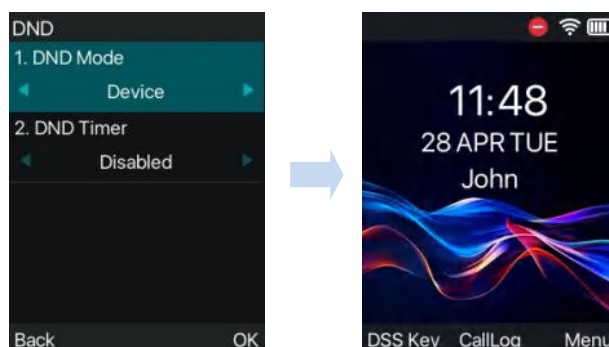
8.10 DND

User may enable Do-Not-Disturb (DND) feature on the device to reject incoming calls (including call waiting). The DND can be enabled on line basis.

Phone user interface enable/disable DND

- Enable/Disable phone all lines DND, the methods as the following:
 - 1) Press **[OK]** key in standby mode to enter menu >> **[Features]** >> **[DND]** to enter the **[DND]** edit page.
 - 2) Use the **[Left/Right navigation keys]** to select the mode as **[Phone]**, and press the **[OK]** key to save after completion.

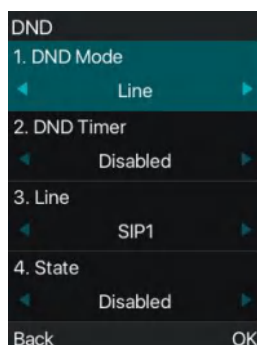
3) The user will see the DND icon  in the status bar when DND is enabled, and the appearance of this icon shows that the device has enabled Do Not Disturb Mode.



Picture 37 - Enable DND

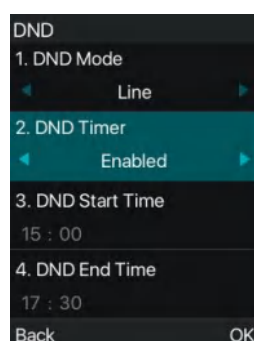
- If the user wants to enable/disable the uninterrupted function on a specific line, the user can set the methods as the following:

- 1) Press **[OK]** key in standby mode to enter Menu >> **[Features]** >> **[DND]** to enter the **[DND]** edit page.
- 2) Use the **[Left/Right navigation keys]** to select the line to adjust the DND mode and status, and press the **[OK]** key to save after completion.
- 3) The user will see the DND icon  in the status bar, and the sip-line has enabled the mode of "DND".



Picture 38 - DND setting interface

- The user can also use the DND timer. After the setting, the DND function will automatically turn on and the DND icon will turn red when ringing.



Picture 39 - DND timer

WEB interface enable/disable DND

- After logging into the web page, enter **[Phone setting] >> [Features] >> [DND settings]**, set the DND type (off, phone, line), and DND timing function.

Picture 40 - DND Settings

The user turns on the DND for a specific route on the web page: Enter **[Line] >> [SIP]**, select a **[Line] >> [Basic settings]**, and enable DND.

Picture 41 - Line DND

8.11 Call Forward

Call forward is also known as 'Call Divert' which is to divert the incoming call to a specific number based on the conditions and configurations. Users can configure the call forward settings of each line. There are three types,

- Unconditional Call Forward** – Forward any incoming call to the configured number.

- **Call Forward on Busy** – When the user is busy, the incoming call will be forwarded to the configured number.
- **Call Forward on No Answer** – When the user does not answer the incoming call after the configured delay time, the incoming call will be forwarded to the configured number.

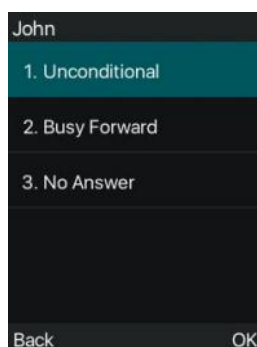
Phone interface: Default standby mode

- 1) Press **[OK]** key in standby mode to enter menu >> **[Features]** >> **[Call Forward]**, select the line by **[Up/Down navigation key]**, press **[OK]** key to set call forward.



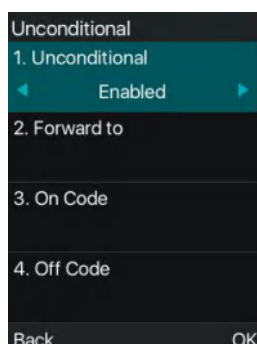
Picture 42 - Select the line to set up call forwarding

- 2) Select the call forward type by pressing the **[Up/Down navigation key]**. Click **[OK]** to configure call forwarding and delay time.



Picture 43 - Select call forward type

- 3) Select enable/disable by pressing the **[Left/Right navigation key]**.



Picture 44 - Enable call forwarding and configure the call forwarding number

- 4) Browse the parameters set by the **[Up/Down navigation key]** and enter the required information.
When finished, press the **[OK]** key to save the changes.

WEB interface: Enter **[Line]** >> **[SIP]**, Select a **[Line]** >> **[Basic settings]**, and set the type, number and time of forward forwarding.

The screenshot shows the 'Basic Settings' for a SIP line. A red box highlights the 'Call Forward' settings, which include:

- Enable Auto Answering: ☐
- Call Forward Unconditional: ☐
- Call Forward on Busy: ☐
- Call Forward on No Answer: ☐
- Call Forward Delay for No Answer: 5 (0~120)second(s)
- Auto Answering Delay: 5 (0~120)second(s)
- Call Forward Number for Unconditional:
- Call Forward Number for Busy:
- Call Forward Number for No Answer:
- Transfer Timeout: 0 second(s)

Other settings visible include Conference Type: Local, Server Conference Number: , Voice Message Number: , and Enable Hotline: ☐.

Picture 45 - Set call forward

8.12 Call Transfer

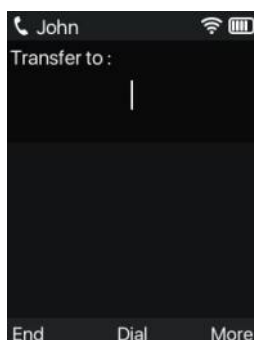
When the user is in an active call and wishes to transfer the call to a third number so that the two parties can communicate directly,, there are three ways to transfer the call, blind transfer, attended transfer and Semi-Attended transfer.

- Blind transfer: No consultation with the third party is required. The call is transferred directly, and the third party will receive the incoming call only after the transfer is completed successfully.
- Semi-Attended transfer: The user first calls the third party. After hearing the ringback tone, then the user transfers the original call to the third party.
- Attended transfer: The user first calls the third party, and transfers the original call to the third party only after the third party answers the incoming call.

Note! For more transfer Settings, please refer to [12.7 Line >> Dial Plan](#)

8.12.1 Blind transfer

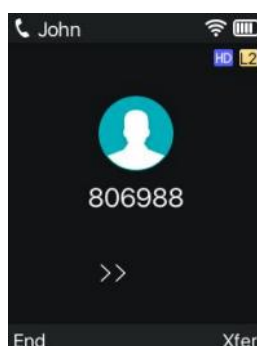
During the call, the user can press **[soft key-Xfer]** or the physical key **[Call transfer key]** ,then enter the number or press the **[Phonebook]** key or the **[CallLog]** key to select the number to transfer, and then press the **[Call transfer key]** again to transfer the call to the third party. When the third party rings successfully, the phone will show that the transfer is successful and hang up automatically.



Picture 46 - Transfer interface

8.12.2 Semi-Attended transfer

During the call, the user can press **[soft key-Xfer]** / **[Call transfer key]** to enter the target transfer number or press the **[Phonebook]** key or the **[CallLog]** key to select the number, and then press the **[Dial key 📞]**. Before the third party answers the call, as long as the call is ringing normally, the user can press the **[transfer]** key / physical **[transfer key]** to make the semi-attendance transfer or press the **[end]** key to cancel the semi-attendance transfer.

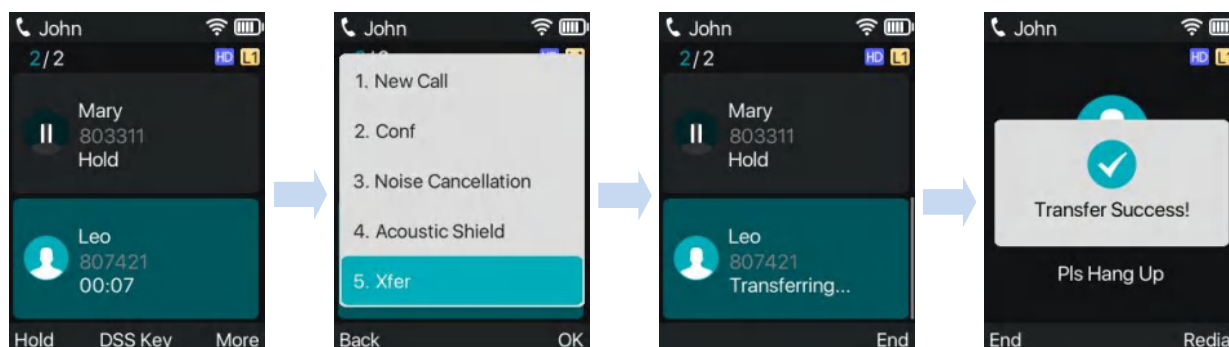


Picture 47 - Semi-Attended transfer

8.12.3 Attended transfer

Attendance transfer is also known as "courtesy mode", which is to transfer the call by calling the other party and waiting for the other party to answer the call.

The same procedure to calling. In dual call mode, press **[soft key-Xfer]** / **[Call transfer key]** to transfer the first call to the second call.



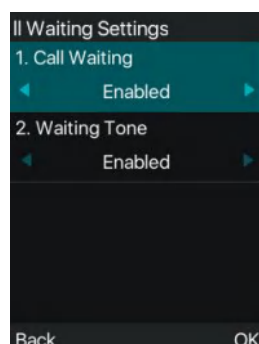
Picture 48 - Attended transfer

8.13 Call Waiting

- Enable call waiting: new calls can be accepted during a call.
- Disable call waiting: new calls will be automatically rejected and a busy tone will be prompted to caller.
- Enable call waiting tone: when you receive a new call on the line, the tone will beep.

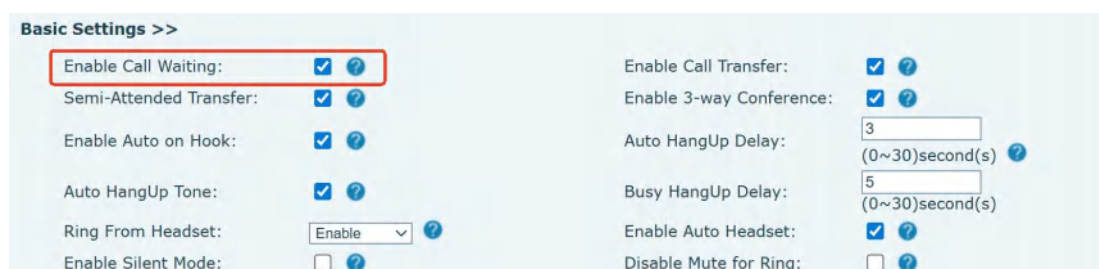
The user can enable/disable the call waiting function in the phone interface and the web interface.

- Phone interface: Press **[OK]** key in standby mode to enter menu >> **[Features]** >> **[Call waiting]**, press the **[Left/Right navigation key]** to enable/disable call waiting and call waiting tone.

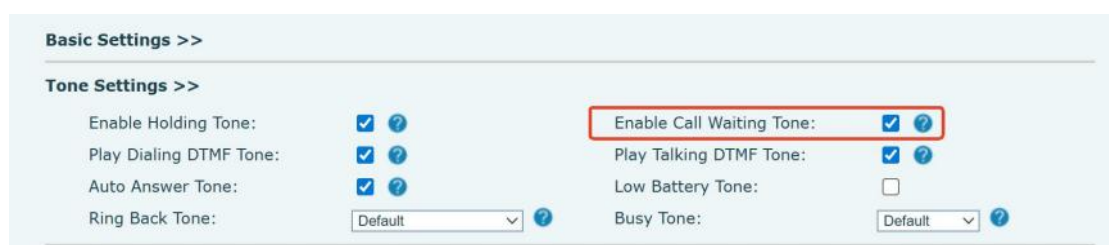


Picture 49 - Call waiting setting

WEB interface: Enter **[Phone Settings]** >> **[Features]** >> **[Basic Settings]**, enable/disable call waiting and call waiting tone.



Picture 50 - Web call waiting setting



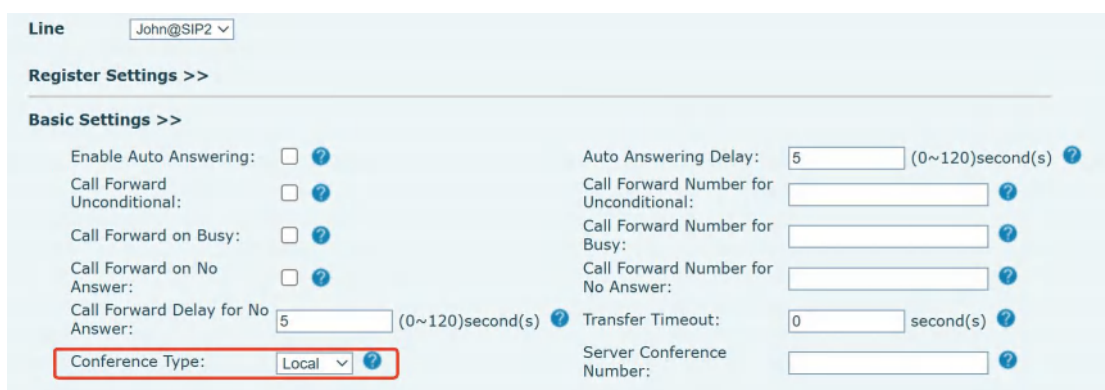
Picture 51 - Web call waiting tone setting

8.14 Conference

8.14.1 Local Conference

To conduct local conference, the user needs to log in the web page and enter **[Line]** >> **[SIP]** >> **[Basic**

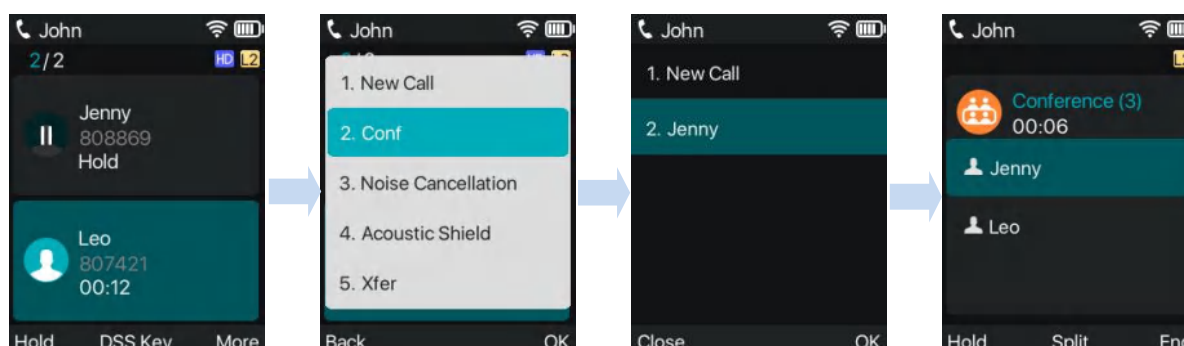
settings]. The meeting mode is set as local (the default is local mode), as shown in the figure:



Picture 52 - Local conference setting

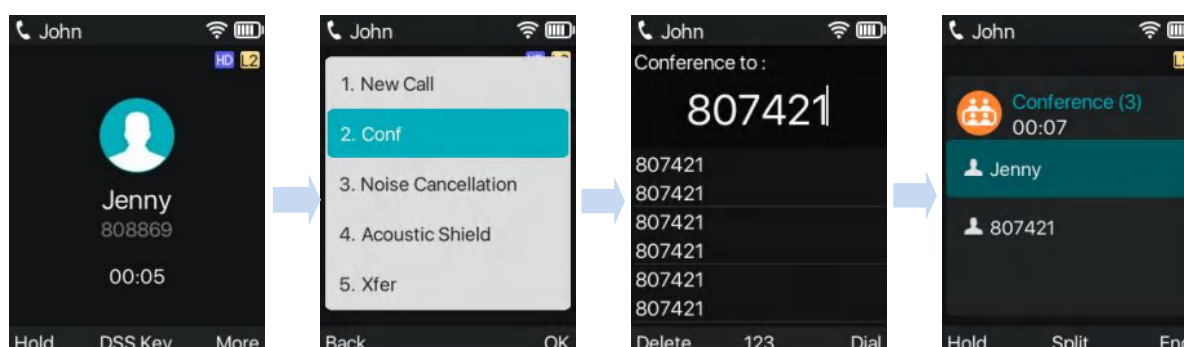
Two ways to create a local conference:

1) The device has two calls, press **[soft key-More]** >> **[Conf]** on the call interface, select another existing number when selecting the conference number, and press the **[OK]** key to establish a local 3-way conference as shown in the figure.



Picture 53 - Local conference (1)

2) If there is only one active call on the device, press **[More]** >> **[Conf]** on the phone interface first. Enter the number to be added to the conference and press **[Dial]**. After the remote party answers, two active calls will be established successfully. Then press **[More]** >> **[Conf]** again to set up a local three-party conference call. As shown in the figure:



Picture 54 - Local conference (2)

Note: During a conference, press the **[Split]** key to separate the conference. The user will continue the call with the currently selected number, while all other parties in the conference will be automatically placed on hold. Press the **[End]** key to terminate the call..

8.14.2 Network Conference

Users need server support for network conference.

Log in the web page, enter **[Line]** >> **[SIP]** >> **[Basic settings]**, set the conference mode as server mode (default is local mode), set the server conference room number (please consult your system administrator), as shown in the figure:

The screenshot shows the 'Basic Settings' page for a SIP line named 'John@SIP2'. The 'Conference Type' is set to 'Local' and the 'Server Conference Number' is empty. These two fields are highlighted with a red rectangle. Other settings include checkboxes for 'Enable Auto Answering', 'Call Forward Unconditional', 'Call Forward on Busy', and 'Call Forward on No Answer'. There are also input fields for 'Auto Answering Delay' (5 seconds), 'Call Forward Number for Unconditional', 'Call Forward Number for Busy', 'Call Forward Number for No Answer', 'Call Forward Delay for No Answer' (5 seconds), and 'Transfer Timeout' (0 seconds).

Picture 55 - Network conference

Method to join a network conference:

- Multi-way call number of network conference room and enter the password then all enter the conference room.
- The two phones have established common calls. Press the conference key to invite new members to the conference. Follow the voice prompt to operate.

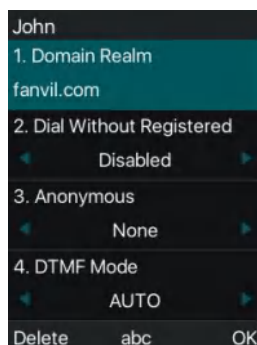
Note: the upper limit of the number of participants in the network conference varies according to the server.

8.15 Anonymous Call

8.15.1 Anonymous Call

The phone can set up anonymous calls to hide the calling number and the calling name.

- Phone interface: press **[OK]** key in standby mode to enter menu >> **[Advanced]** >> **[Accounts]** >> **[SIP Line]** >> **[Advanced]**.
- The default is none, which is off, and RFC3323 and RFC3325 are optional.
- Select any one to open the anonymous call.



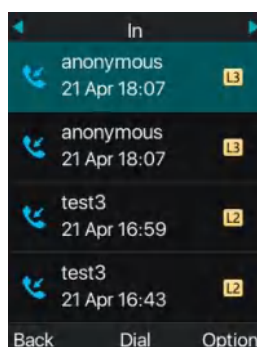
Picture 56 - Enable anonymous call

- On the web page [Line] >> [SIP] >> [System] can also open the mode of anonymous calls.
- Setting to enable anonymous calls also corresponds to the SIP line. That is, the setting under the SIP1 page can only take effect on the SIP1 line.

Enable Session Timer: ☐	Session Timeout: 1800 second(s)
Enable BLF List: ☐	BLF List Number:
Response Single Codec: ☐	BLF Server:
Keep Alive Type: UDP	Keep Alive Interval: 30 second(s)
Keep Authentication: ☐	Blocking Anonymous Call: ☐
RTP Encryption(SRTP): Disabled	Enable OSRTP: ☐
Proxy Require:	
User Agent:	Specific Server Type: COMMON
SIP Version: RFC3261	Anonymous Call Standard: None
Local Port: 5060	Ring Type: Default
Enable user=phone: ☐	Use Tel Call: ☐
Auto TCP: ☐	Enable PRACK: ☐
Enable Rport: ☒	Call-ID Format: Sid@Sip

Picture 57 - Enable Anonymous web page call

The following is a transcript of an anonymous call received by the phone.



Picture 58 - Anonymous call log

8.15.2 Ban Anonymous Call

The device can be set to prohibit anonymous calls, that means anonymous calls to the number will be

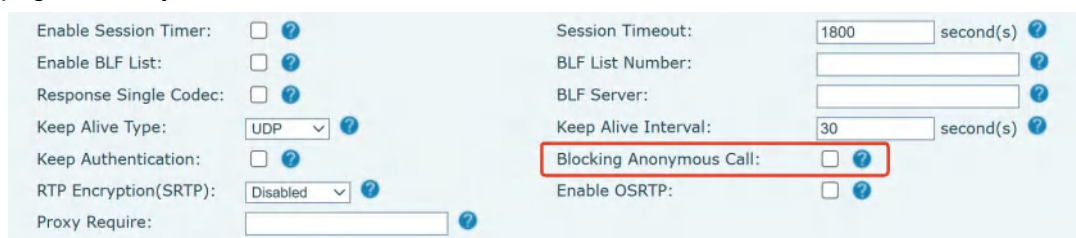
directly rejected.

- Phone interface: press **[OK]** key in standby mode to enter menu >> **[Features]** >> **[Ban anonymous call]**, click to enter and all SIP lines will be displayed.
- Use **[Left/Right navigation key]** to switch the SIP line and enable anonymous call.



Picture 59 - Anonymous calls are not allowed on the phone

- On the web page **[Line]** >> **[SIP]** >> **[Advanced Settings]**, users can also disable anonymous calls.
- The setup to disable anonymous calls also corresponds to the SIP line. That is, the setting under the SIP1 page can only take effect on the SIP1 line.

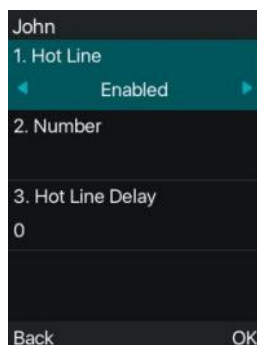


Picture 60 - Page Settings blocking anonymous call

8.16 Hotline

The device supports hotline dialing. After hotline dialing is configured, when you access the dialing interface by pressing the **[Off-hook key]**, **[Hands-free key]** or via the headset, the phone will automatically call the preset hotline number if no dialing operation is performed beyond the set hotline delay time.

- Press the **[OK]** key in the phone's standby mode to enter the menu >> **[Features]** >> **[Advanced]** >> **[Hot Line]**, click to enter and all SIP lines will be displayed.
- Select a SIP line with the **[Up/Down navigation keys]** and press **[OK]** to enter the hotline setting page of the selected SIP line.
- Hotline function is disabled by default. Use the **[Left/Right navigation keys]** to enable or disable the hotline. Enter the desired hotline number and hotline delay time in the input fields below, then press **[OK]** to complete the settings.



Picture 61 - Phone hotline setting interface

- On the website **[Line] >> [SIP] >> [Basic Settings]**, can also set up a hotline.
- The setup hotline also corresponds to the SIP line. That is, the hotline set in the SIP1 web page can only be activated in the SIP1 line.

Basic Settings >>

Enable Auto Answering: ☐ ?	Auto Answering Delay: 5 (0~120)second(s) ?
Call Forward Unconditional: ☐ ?	Call Forward Number for Unconditional: ?
Call Forward on Busy: ☐ ?	Call Forward Number for Busy: ?
Call Forward on No Answer: ☐ ?	Call Forward Number for No Answer: ?
Call Forward Delay for No Answer: 5 (0~120)second(s) ?	Transfer Timeout: 0 second(s) ?
Conference Type: Local ?	Server Conference Number: ?
Subscribe For Voice Message: ☐ ?	Voice Message Number: ?
Voice Message Subscribe Period: 3600 (60~999999)second(s)	Enable Hotline: ☐ ?
Hotline Delay: 0 (0~30)second(s) ?	Hotline Number: ?
Dial Without Registered: ☐ ?	Enable Missed Call Log: ☒ ?
DTMF Type: AUTO ?	DTMF SIP INFO Mode: Send 10/11 ?
Request With Port: ☒ ?	Enable DND: ☐ ?

Picture 62 - Hotline set up on web page

8.17 Emergency Call

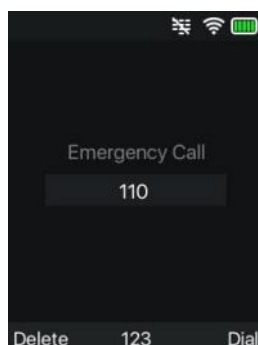
The emergency call function allows the handset to call preset emergency numbers even when the keypad is locked. Users can access emergency call services while the handset is locked.

- 1) Configure the emergency call number: log in the phone page, enter the **[Phone Settings] >> [Features] >> [Basic Settings]** page, set up the emergency call number, if the user need to set up more than one emergency call number, please use ", "to separate.

Allow IP Call:	☒	P2P IP Prefix:	
Caller Name Priority:	LocalContact-NetContact-SIP DisplayName	Emergency Call Number:	110
Search path:	LDAP	LDAP Search:	LDAP 1
Caller Display Type:	Normal	Push XML Server:	
Restrict Active URI Source IP:		Enable Multi Line:	☒
Enable Pre-Dial:	☒	Contact As Allowed List Type:	NONE
Line Display Format:	xxx@SIPn	SIP Notify:	Enable
Block XML When Call:	Enable	Auto Resume Current:	Disable
Call Number Filter:		Ring Timeout:	120
Call Timeout:	120 (1~3600)second(s)	Display BLF PickUp Popup:	☐
Enable Push XML Auth:	☐	Ring Type For BLF PickUp:	Splash
Play BLF PickUp Tone:	☐	Enable Display To Info:	☐
Ring Priority:	Priority		

Picture 63 - Set up an emergency call number

- 2) When the phone set the keyboard lock, the user can call the emergency call number without unlocking, as shown in the figure:



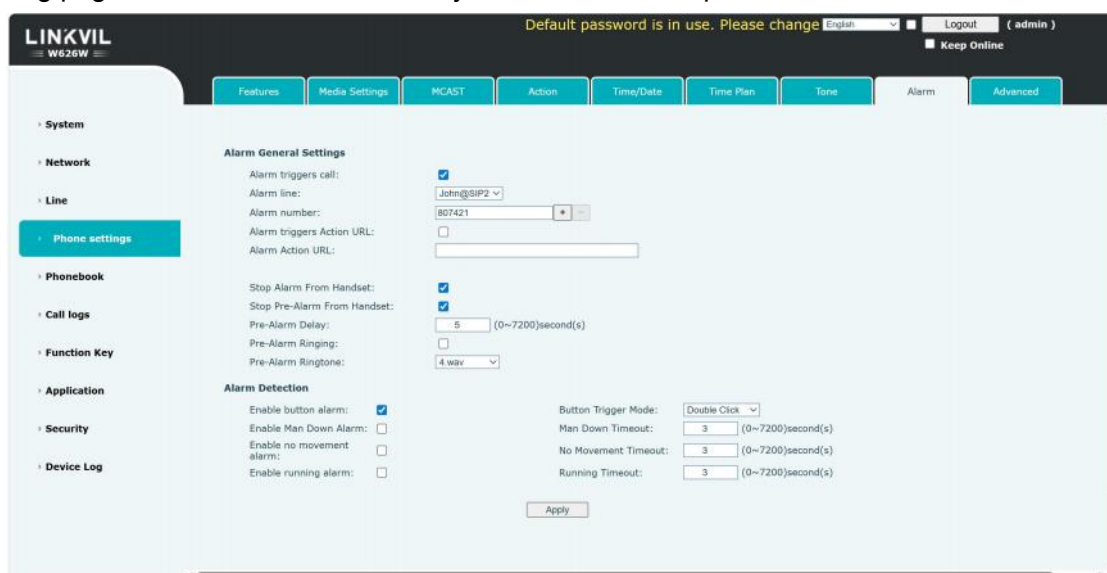
Picture 64 - Dial the emergency number

9 Advance Function

9.1 Alarm

The W626W features a unique alarm function. Under specific conditions, the W626W supports motion detection and automatically performs alarm operations according to preset configurations. This effectively improves the reliability of the handset in industrial scenarios and provides safety protection for users in outdoor operations, emergency rescue and other environments. The alarm function is disabled by default and can be configured on the web page.

Setting Method: Access W626W device's web interface >> **[Phone settings]** → **[Alarm]** to enter the alarm setting page, where the user can modify and save relevant parameters of the alarm function.



Picture 65 - Alarm Settings page on the Web Page

9.1.1 Alarm General Settings

In Alarm General Settings, the user can configure alarm-triggered call and alarm-triggered Action URL for quick alarm information transmission (as shown in the figure below).

Alarm General Settings

Alarm triggers call:

☒

Alarm line:

John@SIP2

Alarm number:

807421

+

-

Alarm triggers Action URL:

☐

Alarm Action URL:

Stop Alarm From Handset:

☒

Stop Pre-Alarm From Handset:

☒

Pre-Alarm Delay:

5

(0~7200)second(s)

Pre-Alarm Ringing:

☐

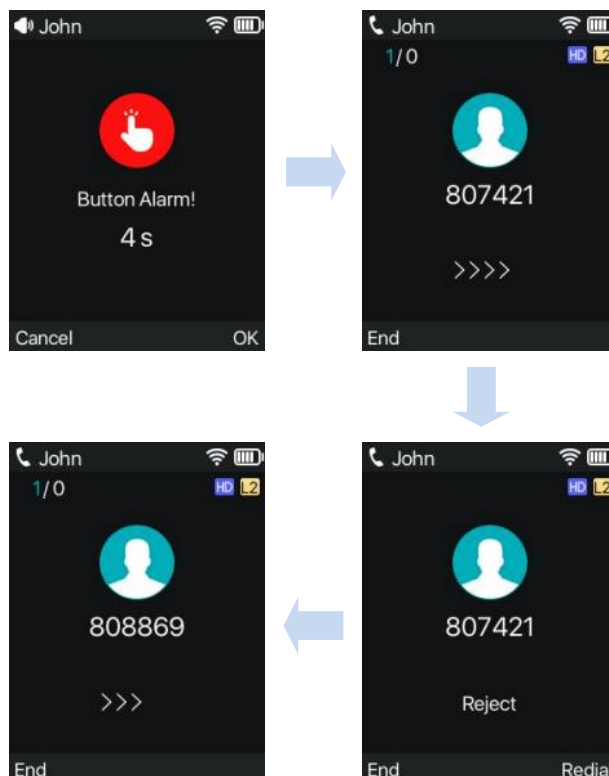
Pre-Alarm Ringtone:

4.wav

Picture 66 - Alarm General Settings

After checking “Alarm-triggered Call”, the user need to set the alarm line and alarm numbers, which correspond to the SIP line for automatic call and the target call number when the alarm is triggered respectively. The default value of the alarm line is “AUTO”, which uses SIP1 by default and can be set to a specific SIP line. Up to 5 alarm numbers can be configured. The user may add or delete alarm numbers via the **[Add +]** / **[Remove -]** buttons next to the alarm number input fields.

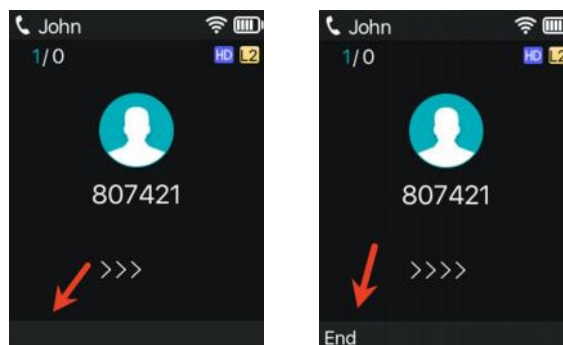
If two or more alarm numbers are added, the handset will dial them in top-to-bottom order when an alarm is triggered. When the preceding number is unanswered or rejected, the handset will automatically call the next number until one call is answered or the last number is dialed.



Picture 67 - Alarm calls

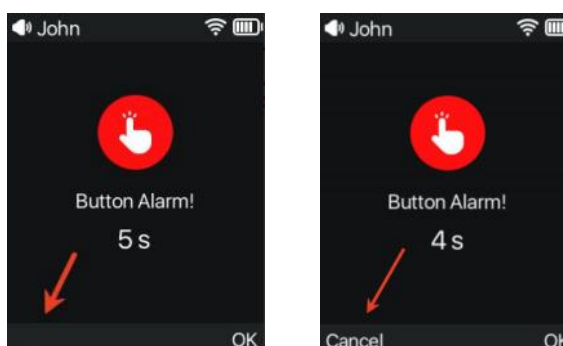
After enabling “Alarm-triggered Action URL”, the user shall enter a valid value in the “Alarm Action URL” input field. When an alarm is triggered on the phone, it will automatically send information according to the configured Action URL.

After checking "Stop alarm call from phone", the user can press the **[End]** key or the **[Hang up key 📞]** on the alarm call interface to cancel the call. Once the call is canceled, the current alarm will terminate simultaneously. If this option is unchecked, the alarm cannot be canceled manually on the phone until the alarm call is answered or dialing to all preset alarm numbers is completed.



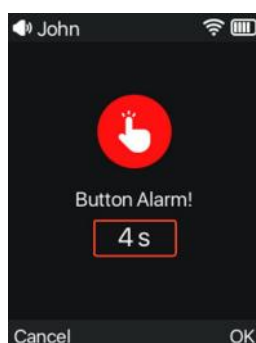
Picture 68 - Cancel the alream call

After checking "Allow pre-alarm cancellation", the user can press the **[Cancel]** key or the **[Hang up key 📞]** on the pre-alarm interface when an alarm is triggered to terminate the current alarm. Once canceled successfully, the alarm process will end immediately. If this option is unchecked, pre-alarm cancellation will be disabled on the handset. The system will proceed with the normal alarm call procedure until the alarm call is answered or dialing to all preset alarm numbers is completed.



Picture 69 - Cancel the pre-alarm

The default value of "Pre-Alarm Delay" is 5, which can be customized by the user. Pre-alarm Delay refers to the initial countdown duration on the pre-alarm interface after the phone triggers an alarm, measured in seconds. A larger value means a longer time interval from alarm trigger to formal alarm execution. When the countdown expires, the phone will automatically initiate alarm calls and send the configured Action URL. Users may also press the **[OK]** key before the countdown ends to trigger the alarm calls and Action URL in advance.



Picture 70 -Pre-alarm delay

After checking "Pre-alarm Ringtone", the phone will play the corresponding pre-alarm prompt tone when an alarm is triggered. The tone will stop automatically once formal alarm operations are executed, and will also terminate early if the alarm is triggered in advance. If "Pre-alarm Ringtone" is unchecked, no pre-alarm prompt tone will be played after alarm triggering. In addition, the pre-alarm tone will remain silent even if this option is enabled when the phone is in mute mode.

9.1.2 Alarm Detection

After enabling the alarm detection for corresponding statuses, the relevant alarm will be automatically triggered once the handset status matches the selected alarm conditions. "Button Alarm" is activated by operating the red alarm key on the top of the handset through double-pressing or long-pressing the key for 3 seconds. All other alarm triggers are based on user motion statuses: fall detection, inactivity, and running detection. The alarm delay time can be configured individually for each trigger mode.

Picture 71 - Alarm detection settings

9.2 Video Preview

W626W supports video preview and video calls with a maximum video resolution of 4CIF, and the default value is CIF. When the video preview function is enabled, the W626W can preview the caller's video content before answering when another device initiates a video call to it.

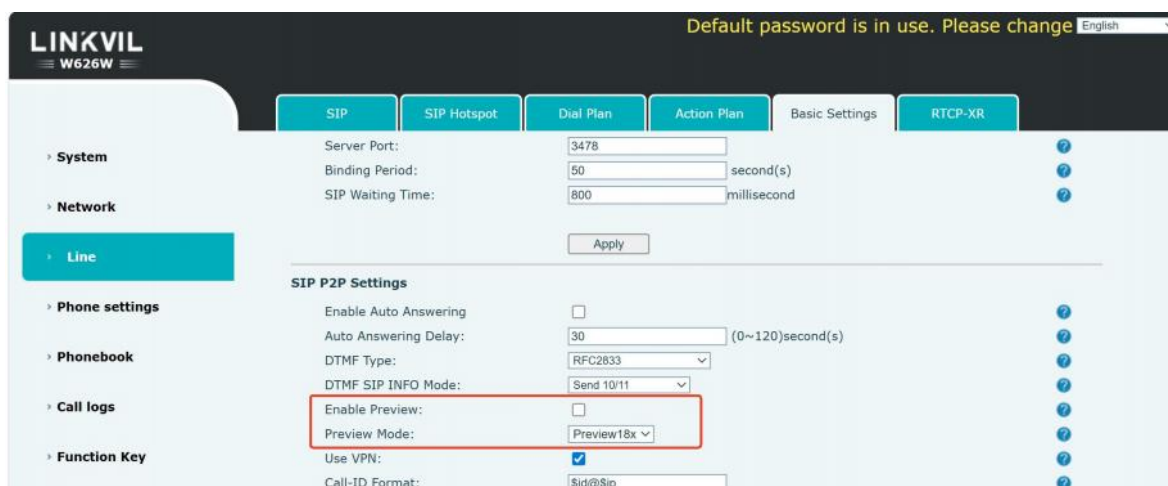
Setting Method:

- Video preview for SIP lines: Access W626W device's web interface >> **[Line]** >> **[SIP]** >> **[Advanced Settings]**, check to **[enable preview]**, and select the preview mode.

Picture 72 - Web interface settings for SIP line video preview

- Setting up video preview for IP calls: Access W626W device's web interface >> **[Line]** >> **[Basic**

Settings] >> **[SIP P2P Settings]**, check to **[enable preview]**, and select the preview mode.



Picture 73 - Video preview settings for IP calls on the web interface

Preview Mode:

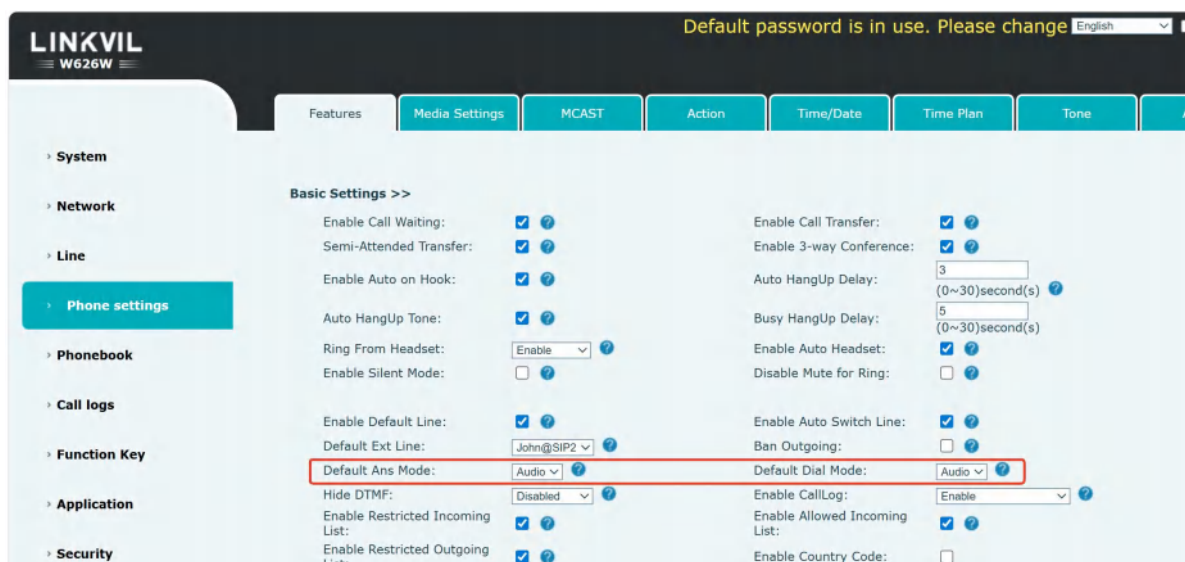
- Preview2xx: Default value. Preview is performed by replying with a 2XX message. During preview, the other end displays the call timing interface, while the local end displays the video preview interface, and the voice call is not connected.
- Preview18x: Preview is performed via 18X messages and requires SIP server support.

9.3 Video Call

After setting the value of **[Default Ans Mode]** / **[Default Dial Mode]** to **[Video]**, when the device answers a call from or makes a call to a video-enabled device, the video interface of the other party will be displayed after the call is connected.

Setting method: Access W626W device's web interface >> **[Phone settings]** >> **[Features]** >> **[Basic Settings]**

- Default Ans Mode:
 - Video: Default value. When the device receives an incoming video call from the other party, it initiates a video call after answering.
 - Audio: When the device receives an incoming video call from the other party, it initiates a voice call after answering.
- Default Dial Mode:
 - Video: Call the other party's device in video mode. If the other party's device also answers in video mode, a video call will be initiated after the call is answered.
 - Audio: Default value. Call the other party's device in voice mode, and a voice call will be initiated after the call is answered.

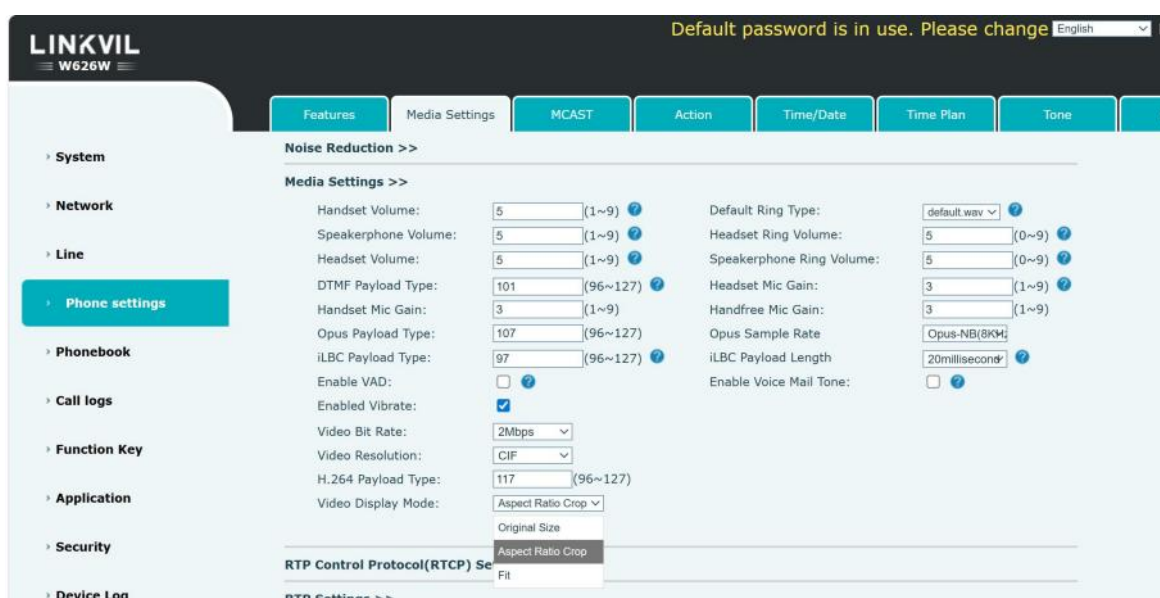


Picture 74 - Web interface settings for default ans/dial mode

- Video Display Mode

Setting method: Access W626W device's web interface >> **【Phone Settings】** >> **【Media Settings】** >> **【Media Settings】** >> **【Video Display Mode】**

- Aspect Ratio Crop: Current setting. Crops the image in equal proportion to enable full-screen video display with no distortion (default value).
- Fit: Compresses the original video in equal proportion and fills the blank areas with black, ensuring no distortion.
- Original Size: Displays the original video in full screen, which may result in stretching distortion.



Picture 75 - Web interface settings for video display mode

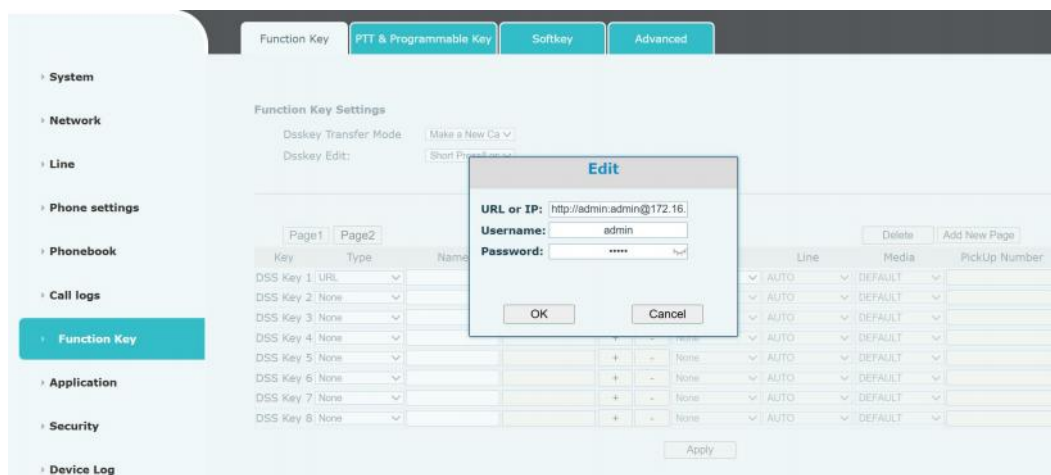
9.4 One-Click to Open Door

The device supports one-touch door opening for Fanvil access control via function keys both in standby mode and during a call. W626W detects that the peer end complies with the number rules in the access control list, and dynamically displays **[Soft key - Open]**, which is displayed on the far left by default.

9.4.1 Door Opening in Standby Mode

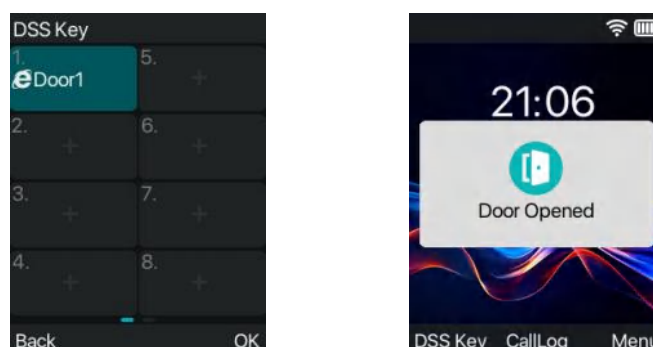
Setting Method: Set the function key type to URL >> Open Door, edit the value of the function key, and save:

- URL or IP: Set the IP address of the access control device.
- Username: Set the username for accessing the access control web interface.
- Password: Set the password for accessing the access control web interface.
- When setting via a configuration file or LCD, the value of the function key can be set to:
http://[Username]:[Password]@[Access Control IP]/cgi-bin/ConfigManApp.com?key=F_LOCK&code=*[Access Control Password]



Picture 76 - Web interface settings for URL door opening

Operation: Press this **[DSS Key]** to open the door when in standby/call mode.

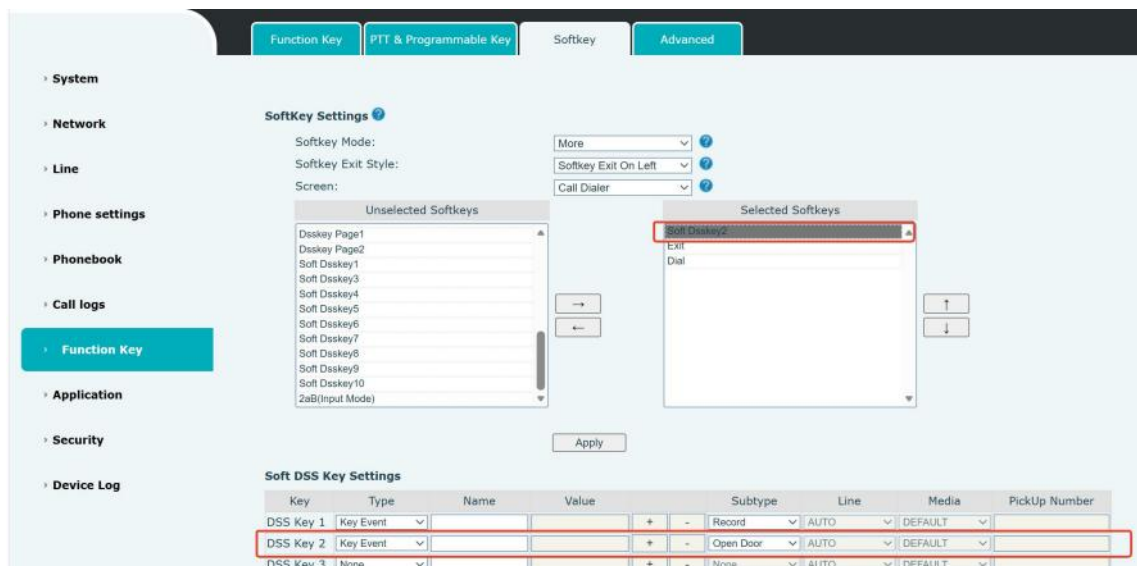


Picture 77 - Door opening in standby mode

9.4.2 Door Opening in Standby Mode

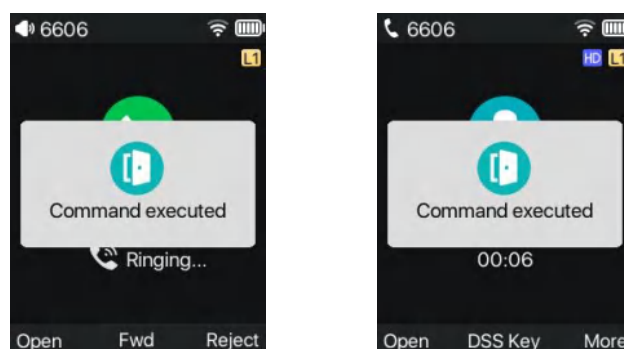
Setting Method: Access W626W device web page >> **[Function Key]** >> **[Softkey]**, select the display

page as "Ringing", "Voice Call" or "Video Call", and move SoftDsskey2 (which has been set as the default Type of Open Door) to the "Selected Softkeys" area on the right.



Picture 78 - Web interface settings for door-opening softkey

Operation: When there is an incoming call from the Fanvil access control device, press the [soft key-Open] on the W626W during ringing or a call to open the door with one touch.



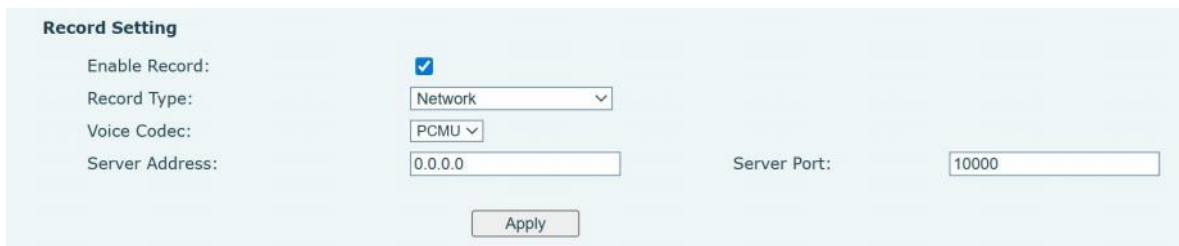
Picture 79 - Door opening via softkey on the handset

9.5 Record

The device supports recording during a call.

9.5.1 Server Record

When using the network server to record, it is necessary to open the recording in the phone web page [Application] >> [Manage recording]. The type is selected as network, and the address and port of the recording server are filled in and the voice coding is selected. The web is as follows:

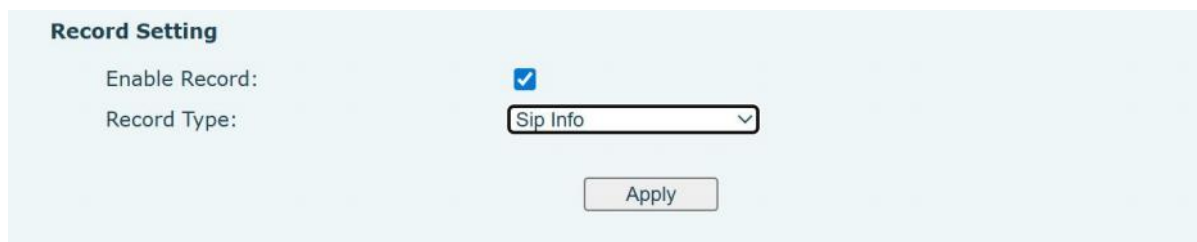


Picture 80 - Web server recording

Note: to be used with Fanvil recording software.

9.5.2 SIP INFO Record

The phone is registered with a server that supports SIP INFO recording. After registering the account, check the recording module of **[Application] >> [Record Setting]** to open the recording, and the record type is SIP INFO.



Picture 81 - Web SIP info recording

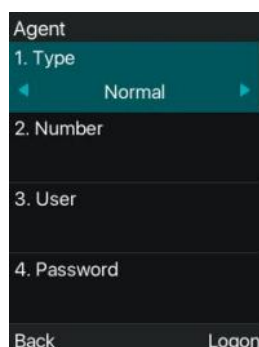
9.6 Agent

Agent (Agent function) of the phone can be realized: when multiple people use a device for Agent services at different times, he or she can quickly register his or her SIP account on the same server. The Agent functions of the phone can be divided into Normal and Hotel Guest. The Hotel Guest mode requires server support.

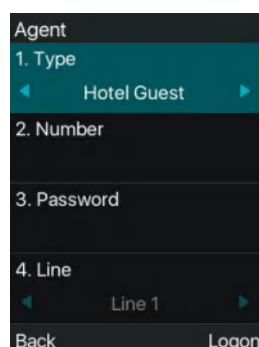
Normal Mode:

Configure agent function: set a DSS Key as agent, press the function key or press **[OK]** key in the phone's standby mode to enter menu >> **[Features] >> [Agent]** to enter the agent page. The SIP server needs to be configured before the account can be configured.

Note: The password to enter the proxy configuration is the same as the menu password, the default is 123.



Picture 82 - Configure the agent account in normal mode



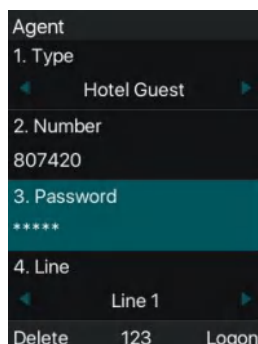
Picture 83 - Configure the proxy account-hotel Guest mode

Table 7 - Agency mode

Parameter	Description
Normal mode	
Number	Set the proxy account number.
User	Set the proxy account number to verify the user name.
Password	Set the proxy account number to verify the password.
Line	Select the SIP line.
Calllog	Users can choose to save all types, or delete
Hotel Guest mode	
Number	Set the proxy account number.
User	Set the proxy account number to verify the user name.
Password	Set the proxy account number to verify the password.
Line	Select the SIP line.
CallLog	Users can choose to save all types, or delete.
Status	The user can select the status of the number, including Login, Logout, Unavailable, Available, and Wrap-up

Using agent functions:

- 1) When the phone has been configured on SIP server, fill in the correct number and user name password, click login and then the phone can be registered to the SIP server;
- 2) After registration, click logout and the phone can delete the user name and password, and log out of the SIP account.
- 3) Click Unregister and the phone retains the user name and password, and logs out of the SIP account.



Picture 84 - Agent login page

9.7 Intercom

When the Intercom is enabled, it can automatically receive calls from the intercom. Access the settings by logging in to the phone web page >> **[Phone Settings]** >> **[Features]** >> **[Intercom Settings]**. Expand the function list of intercom settings and configure parameters as required.



Picture 85 - Web Intercom configure

Table 8 - Intercom configure

Parameter	Description
Enable Intercom	When intercom is enabled, the device will accept the incoming call request with a SIP header of Alert-Info instruction to automatically answer the call after specific delay.
Enable Intercom Mute	Enable mute mode during the intercom call.
Enable Intercom	If the incoming call is intercom call, the phone plays the intercom tone.

Tone	
Enable Intercom Barge	Enable Intercom Barge by selecting it, the phone auto answers the intercom call during a call. If the current call is intercom call, the phone will reject the second intercom call.

9.8 MCAST

Using the multicast function, voice broadcasts can be easily and conveniently sent to all members within the multicast group. By setting the multicast key on the phone, the multicast RTP stream is sent to the pre-configured multicast address. By configuring the listening multicast address on the phone, the RTP stream of that multicast address is listened to and played.

Picture 86 - Multicast Settings Page

Table 9 - MCAST Parameters on Web

Parameters	Description
Normal Call Priority	Define the priority of the active call. 1-10: Define the priority of normal calls, 1 is the highest priority, 10 is the lowest.
Enable Page Priority	Page priority determines how the phone handles newly received multicast RTP streams during a multicast session. When the page priority switch is enabled, the device automatically ignores lower-priority multicast RTP streams, receives higher-priority ones, and places the current multicast session in hold mode; if disabled, it automatically ignores all received multicast RTP streams.
Name	Listened multicast server name.
Host:port	Listened multicast server's multicast IP address and port.

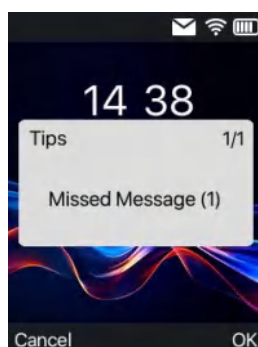
Multicast:

- Go to web page of **[Function Key]** >> **[Function Key]** / **[PTT Key]** / **[Softkey]**, select the type to multicast, set the multicast address, and select the code.
- Click Apply.
- Set up the name, host and port of the receiving multicast on the web page of **[Phone Settings]** >> **[MCAST]**.
- Press the DSS Key of Multicast Key which you set.
- Receive end will receive multicast call and play multicast automatically.

9.9 Message

9.9.1 SMS

If the service of the line supports the function of the short message, when the other end sends a text message to the number, the user will receive the notification of the short message and display the icon of the new SMS on the standby screen interface.



Picture 87 - SMS icon

Send messages:

- Press **[OK]** key in phone's standby mode to enter menu >> **[Message]** >> **[SMS]**.
- Users can create new messages, select lines and send numbers.
- After editing is completed, click **[More]** >> **[Send]**.

View SMS:

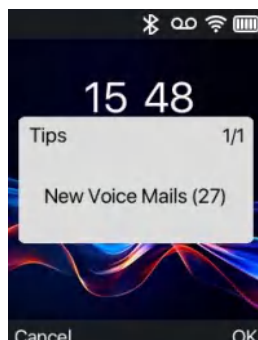
- Press **[OK]** key in phone's standby mode to enter menu >> **[Message]** >> **[SMS]**.
- Select **[Inbox]** to enter the SMS inbox interface.
- Select the unread message and press **[OK]** to read the unread message.

Reply to SMS:

- Press **[OK]** key in phone's standby mode to enter menu >> **[Message]** >> **[SMS]**.
- Select **[Inbox]** to enter the SMS inbox interface.
- Select the message you want to reply to, select **[Reply]**, edit it, and click **[More]** >> **[Send]**.

9.9.2 MWI (Message Waiting Indicator)

If the service of the lines supports voice message feature, when the user is not available to answer the call, the caller can leave a voice message on the server to the user. User will receive voice message notification from the server and device will prompt a voice message waiting icon on the standby screen.



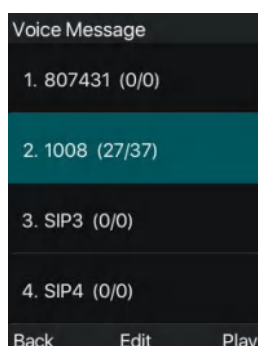
Picture 88 - New Voice Message Notification

Voice message icon

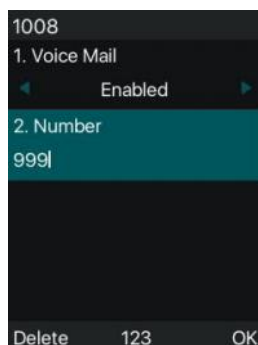
To listen to a voice message, the user must first configure the voicemail number. After the voicemail number is configured, the user can retrieve the voicemail of the default line.

When the phone is in the default standby state,

- Press **[OK]** key to enter menu >> **[Message]** >> **[Voice Message]**
- Open the voice message configuration interface, and select the line to be configured by pressing the Up/Down navigation keys.
- Press the **[Edit]** key to edit the voice message number. When finished, press the **[OK]** key to save the configuration.
- SIP1(m/n), “m” in front of Fanvil line brackets represents unread voice messages, and “n” represents the total number of voice messages.



Picture 89 - Voice message interface



Picture 90 - Configure voicemail number

9.10 SIP Hotspot

SIP Hotspot is a simple and practical function. With simple configurations, the SIP Hotspot function can implement group ringing. SIP accounts can be expanded.

The users can set functions as a SIP Hotspot and other phones set (B and C) function as SIP Hotspot clients. When somebody calls phone set A, phone sets A, B, and C all ring at the same time. When any phone set answers the call, other phone sets stop ringing. The call can be answered by only one phone set. When B or C initiates a call, the SIP number registered by phone set A is the calling number.

To set a SIP Hotspot, please register at least one SIP account.

Picture 91 - Register SIP account

Table 10 - SIP Hotspot Parameters

Parameters	Description
Enable Hotspot	Set it to be Enable to enable the feature.
Mode	Choose Hotspot, phone will be a "SIP Hotspot server"; Choose Client, phone will be a "SIP Hotspot Client"
Monitor Type	Either the Multicast or Broadcast is ok. If you want to limit the broadcast packets, you'd better use broadcast. But, if client choose broadcast, the SIP Hotspot phone

	must be broadcast.
Monitor Address	The address of broadcast, hotspot server and Hotspot client must be same.
Remote Port	Type the Remote port number.
Name	Fill in the name of the SIP Hotspot. This configuration is used to distinguish different hotspots in the network and avoid connection conflicts.
Line Settings	Set whether to associate and enable the SIP Hotspot function on the corresponding SIP line.

Configure SIP Hotspot server:

Picture 92 - SIP Hotspot server configuration

Configure SIP Hotspot client:

To set as a SIP hotspot client, no SIP account needs to be set. The Phone set will automatically obtain and configure a SIP account. On the SIP Hotspot tab page, set Mode to Client. The values of other options are the same as those of the Hotspot.

Picture 93 - SIP Hotspot client configuration

As the Hotspot server, the default extension number is 0. When the phone is used as the client, the extension number is increased from 1, you can view the extension number through the **[SIP Hotspot]** page.

Call extension number:

- The Hotspot server and the client can dial each other through the extension number.
- For example, extension 1 dials extension 0.

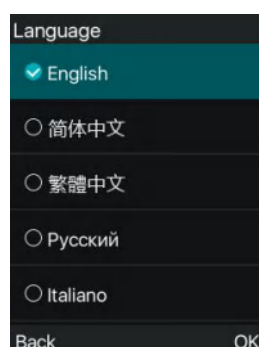
10 Phone Settings

10.1 Basic Settings

10.1.1 Language

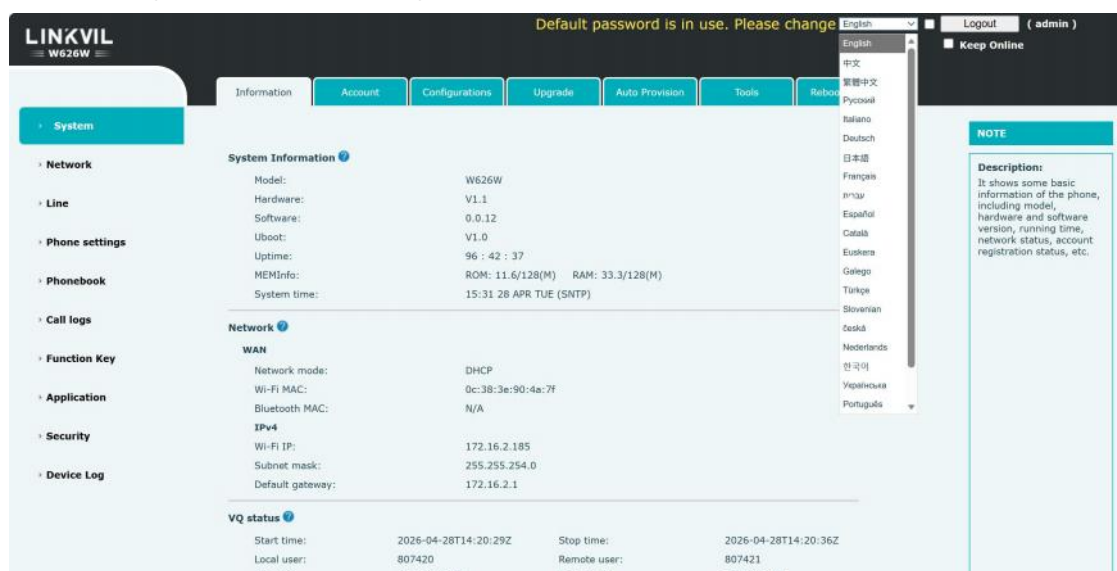
The user can set the phone language through the phone interface and web interface.

- Phone interface: After resetting the factory settings, the user needs to set the language; when setting the language during standby, press **[OK]** key to enter menu >> **[Basic]** >> **[Language]** Settings, as shown in the figure.



Picture 94 - Phone language setting

- Web interface: Log in to the phone web page and set the language in the drop-down box at the top right corner of the page, as shown in the figure:



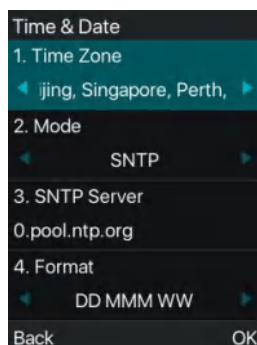
Picture 95 - Language setting on Web page

- On the web interface, there is a checkbox on the right side of the language selection box, labeled “Synchronize Language to Handset”. When checked, the handset system language will be synchronized with the language set on the web page; when unchecked, no synchronization will be performed.

10.1.2 Time & Date

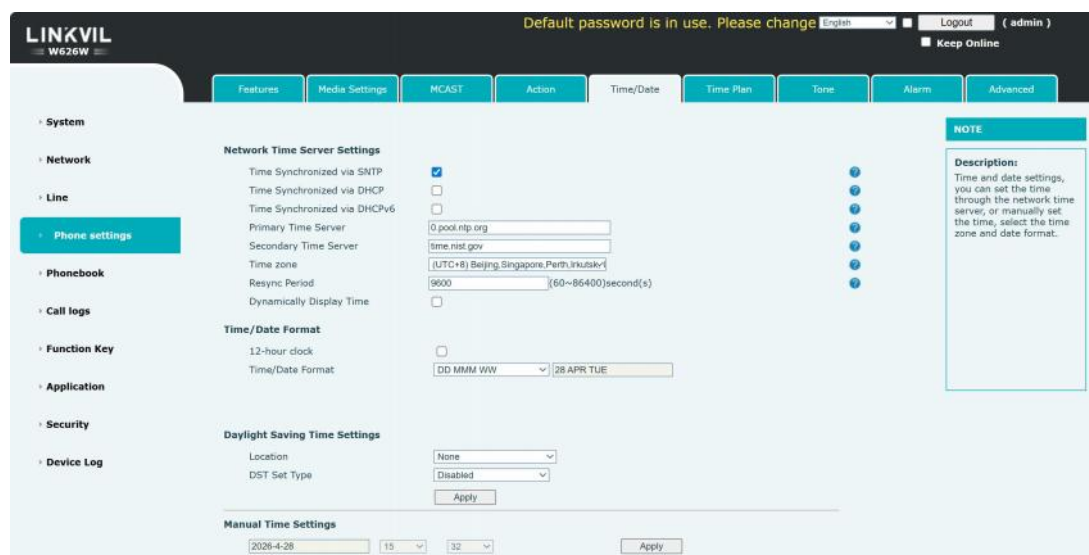
Users can set the phone time through the phone interface and web interface.

- Phone interface: When the phone is in the default standby state, press **[OK]** key to enter menu **>> [Basic] >> [Time & Date]**, use the **[Up/Down navigation key]** to edit parameters, press the **[OK]** to save after completion, as shown in the figure:



Picture 96 - Set time & date on phone

- Web end: Log in to the phone web page and enter **[Phone Settings] >> [Time/Date]**, as shown in the figure:



Picture 97 - Set time & date on web page

Table 11 - Time Settings Parameters

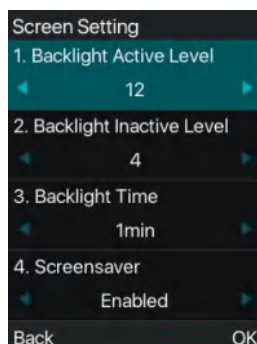
Parameters	Description
Mode	Auto/Manual Auto: Enable network time synchronization via SNTP protocol, default enabled. Manual: User can modify data manually.
SNTP Server	SNTP server address
Time zone	Select the time zone

Time format	Select time format from one of the followings: ■ DD MMM WW ■ WW DD MMM ■ WW MMM DD ■ DD MM YY ■ DD MM YYYY ■ MM DD YY ■ MM DD YYYY
Separator	Choose the separator between year and moth and day
12-Hour Clock	Display the clock in 12-hour format
Daylight Saving Time	Enable or Disable the Daylight Saving Time

10.1.3 Screen

The user can set the phone screen parameters through both of the phone interface and web interface.

- Phone: When the phone is in the default standby state, press **[OK]** key to enter menu >> **[Basic]** >> **[Screen]** to edit the screen parameters. After editing, click **[OK]** to save, as shown in the figure:

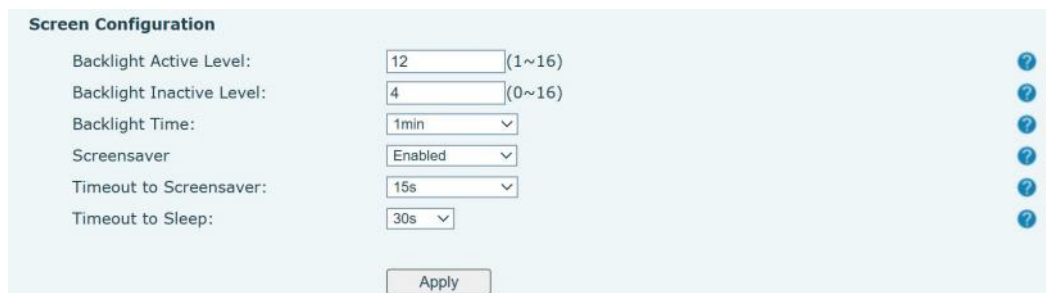


Picture 98 - Set screen parameters on phone

- Web : Go to **[Phone Settings]** >> **[Advanced]**, edit the screen parameters, and click **[Apply]** to save.

10.1.3.1 Brightness and backlight

- Set the brightness level in use from 1 to 16, press **[<]** or **[>]** switch brightness level.
- Set the brightness level in the energy-saving mode from 0 to 16, press **[<]** or **[>]** switch the brightness level.
- Set the backlight time to 1 minute by default. You can turn it off or select 15 seconds / 30 seconds / 1 minute / 2 minutes / 5 minutes / 10 minutes / 30 minutes / 1 hour / 2 hours / 3 hours / 6 hours / 15 hours / Customer Value.
- Web interface: enter **[Phone Settings]** >> **[Advanced]**, edit screen parameters, and click submit to save.



Picture 99 - Page screen Settings

10.1.3.2 Screen Saver

- Press the **[OK]** key to enter the menu >> **[Basic]** >> **[Screen]**, then select **[Screensaver]**. Use the **[Left/Right navigation keys]** to enable / disable the screensaver, and configure the timeout to screensaver, with a default value of 15 seconds. Press the **[OK]** key to save the settings after completion.
- After saving, return to standby mode and enter the screen saver after 15 s, as follows:



Picture 100 - Phone screen saver

10.1.4 Ring

- When the device is in the default standby mode, press **[OK]** key to enter menu >> **[Basic]** >> **[Ring]** to access the ringtone configuration page.
- Find **[Headset]** or **[Hands-free]** item, press **[Left/Right navigator keys]** to adjust the ring volume, save the adjustment by pressing **[OK]** when done.
- Enter **[Ring type]** item, press **[Left/Right navigator keys]** to change the ring type, save the adjustment by pressing **[OK]** when done.

10.1.5 Voice Volume

- When the device is in the default standby mode, press **[OK]** key to enter menu >> **[Basic]** >> **[Voice Volume]** to access the voice volume setting page.
- Find **[Headset]**, **[Hands-free]** and **[Handset]** item, press **[Left/Right navigator keys]** to adjust the audio volume for different mode.
- Save the adjustment by pressing **[OK]** when done.

10.1.6 Greeting Words

- When the device is in the default standby mode, press **[OK]** key to enter menu >> **[Basic]** >> **[Greeting Words]** to open the greeting words setting page.
- Enter the customized greeting words, then save the adjustment by pressing **[OK]** when done.

***NOTICE!** The greeting words can only be displayed in the upper left corner of standby mode when the default option is disabled.*

10.1.7 Reboot

(1) Method 1:

- When the device is in the default standby mode, press **[OK]** key to enter menu >> **[Basic]**
- Find **[Reboot System]** and then press **[OK]**, a prompt message "restart now?" prompts the user.
- Press **[OK]** to restart the phone or **[Cancel]** to exit without restarting.

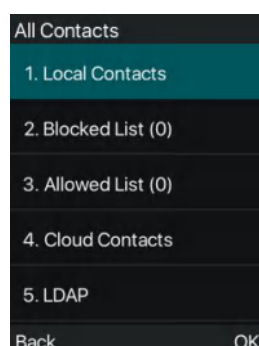
(2) Method 2:

- Users can set the **[OK]** key as a restart key. Configuration steps: Log in to the W626W web page >> **[Function Keys]** >> **[Advanced]** >> **[Programmable Key Settings]**, and set the desktop function of the **[OK]** key to "Reboot". If the **[OK]** key is set to "Reboot", it can no longer be used to enter the menu. To retain the menu access function via the **[OK]** key, set its desktop function to "Main Menu" on this page.
- When the setting is done, press **[OK]**, a prompt message "restart now" prompts the user.
- Press **[OK]** to restart the phone or **[Cancel]** to exit without restarting.

10.2 Phone Book

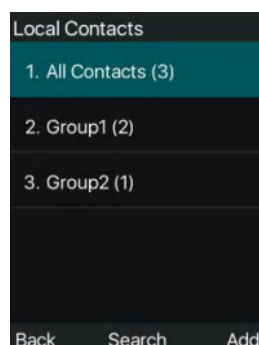
10.2.1 Local Contact

User can save contacts' information in the phone book and dial the contact's phone number(s) from the phone book. To open the phone book, user should press **[Menu]** >> **[Contact]** in the default standby screen or keypad. By default the phone book is empty, user may add contact(s) into the phone book manually or from call logs.



Picture 101 - Phone book screen

***Note!** Phone user account can store contact information, different models and specifications.*



Picture 102 - Local Phone book

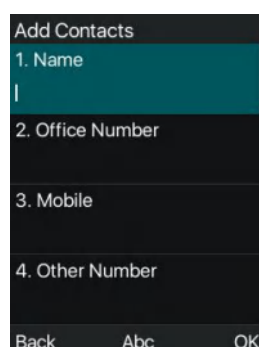
When there are contact records in the phone book, the contact records will be arranged in the alphabet order. User may browse the contacts with **[Up/Down navigator keys]**. The record indicator tells user which contact is currently focused. User may check the contact's information by pressing **[OK]** key.

10.2.1.1 Add / Edit / Delete Contact

To add a contact, press the **[OK]** key to enter Menu >> **[Contact]** >> **[Local Contacts]**, then select a group or create a new group to access the contact group page. If the group contains no contacts, press **[Add]** to open the contact creation page, fill in the relevant information, and press **[OK]** to save.

If the group already has existing contacts, press **[Option]** >> **[New Contact]** to enter the editing page, complete the required fields, and press **[OK]** to finish adding the new contact. The information required on the new contact adding page is as follows:

- Contact Name
- Tel. Number
- Mobile Number
- Other Number
- Line
- Ring Tone
- Contact Group



Picture 103 - Add New Contact

On the contact group page, users can edit the selected contact by pressing **[Option]** >> **[Edit]**. To delete

a contact, select the target contact on the contact group page, press **[Option]** >> **[Delete]**, and confirm with **[OK]** to complete the deletion.

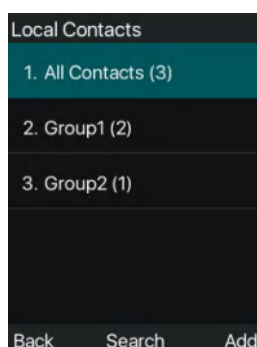
10.2.1.2 Add / Edit / Delete Group

By default, the group list is blank. The user can create his/her own groups, edit the group name, add or remove contacts in the group, and delete a group.

To manage contact groups, press the **[OK]** key to enter Menu >> **[Contact]** >> **[Local Contacts]** to access the local contacts page, then perform operations on each group:

- To add a group, press **[Add]** key, enter the group name and select the ring type, then press **[OK]** to complete group creation.
- To delete a group, press **[Option]** >> **[Delete]** key.
- To edit a group, press **[Edit]** key.

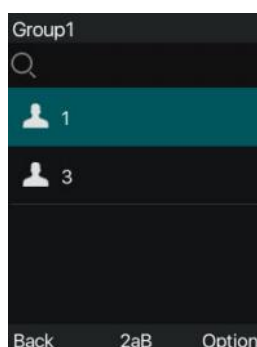
The Number behind the group name means the total contacts number of selected groups.



Picture 104 - Group List

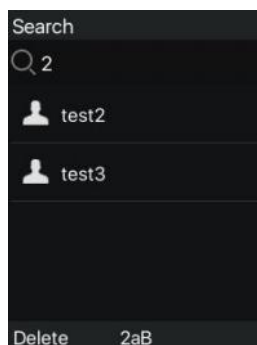
10.2.1.3 Browse and Add / Remove Contacts in Group

Open the group list. Users can select a contact by pressing the **[Up/Down navigation keys]**, then press the **[OK]** key to view the contact's detailed information.



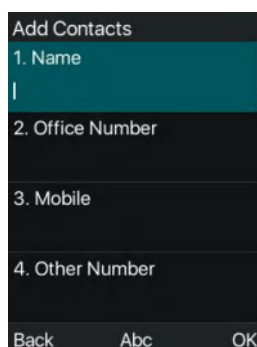
Picture 105 - Browsing Contacts in a Group

When the user browses the contacts in the group, press the keyboard to enter the characters, which will trigger the search function and display the fuzzy search matching contact list.



Picture 106 - Search for existing contacts

When user is browsing contacts of a group, user can also add contacts in that group by pressing **[Add]** key to enter the group contacts management interface, then press **[OK]** key to save the contact. The contact will also be added in local contacts. User can delete contact from group by **[Option] >> [Delete]**.



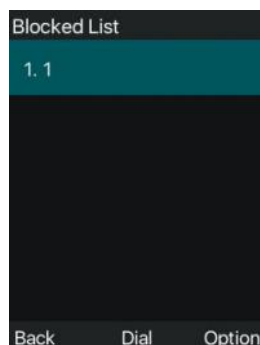
Picture 107 - Add Contacts in a Group

10.2.2 Blocked List

The phone supports a blocked call number list. When a number is added to this list, incoming calls from it will be rejected automatically, and a missed call record will be displayed on the phone. While incoming calls from blocked numbers are declined, the phone can actively initiate outgoing calls to these numbers normally.

Multiple methods are available to add numbers to the blocked call list on the phone.

- The first method: In standby mode, press the **[OK]** key to enter Menu >> **[Contact]** >> **[Blocked List]**. If the list is empty, press **[Add]** to create a new blocked entry. If there are existing entries in the list, select any contact, then press **[Option] >> [New Contact]** to add a new blocked number.
- The second method: Select any number in the phone book (local or network contacts). Press **[Option] >> [Add to Blocked List]** to add the number to the blocked list.
- The third method: Select any number in the call log, press **[Option] >> [Add to Blocked List]** to add the number to the list.



Picture 108 - Add Blocked List

There are various ways to add number to the Blocked List on web page.

- The first method is to log in to the handset web page >> **[Phonebook]** >> **[Call List]** >> **[Restricted Incoming Calls]** to add entries.
- The second method: On the handset web page >> **[Phonebook]** >> **[Contacts]** >> **[Contact List]** or **[Phone Book]** >> **[Cloud phonebook]**. Check one or multiple numbers, then click **[Add to Blocked List]** to complete the addition.
- The third method: In the handset web page >> **[Call logs]**, check one or more numbers, and click **[Add to Blocked List]** to complete the addition.



Picture 109 - Web Blocked List

10.2.3 Cloud Contacts

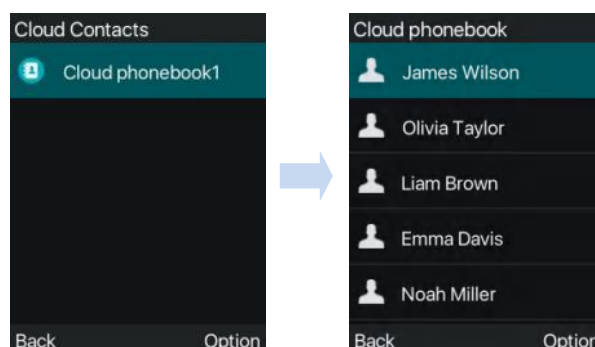
10.2.3.1 Configure Cloud Contacts

The cloud phone book allows users to download the phone book from the cloud server to the handle locally. In the scenario of using the phone book in the office, it is very convenient for users to use the cloud phone book. The phone book can be downloaded to the local and used at one click from the server of the cloud phone book, which is very convenient when creating and maintaining the contact list.

NOTICE! In order to ensure that the content in the cloud phone book is the latest version, each time the cloud phone book is opened on the handle, the contacts in the cloud phone book are only temporarily downloaded to the local. The download speed depends on the quality of the network connected when the handle is used, and it usually takes a few seconds to complete the download. Therefore, in order to save the time that users spend waiting to download the cloud phone book, we strongly recommend that users save important contact information from the cloud phone book to the local.

To open the cloud phonebook list on the phone, press the **[OK]** key on the standby screen to enter Menu >> **[Contact]** >> **[Cloud Contacts]** for viewing.

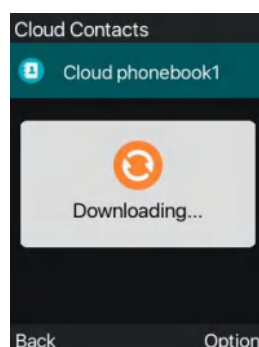
TIPS! The first configuration on cloud phone should be completed on Web page by selecting [Phonebook] >> [Cloud phonebook]. The setting of addition/deletion on device could be done after the first setting on Web page.



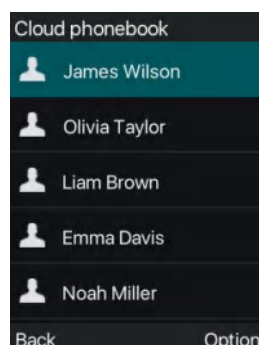
Picture 110 - Cloud phone book list

10.2.3.2 Downloading Cloud Contacts

In cloud contacts screen, user can open a cloud contact by pressing [OK] / [Enter] key. The device will start downloading the phone book. The user will be prompted with a warning message if the download fails. Once the cloud phone book is downloaded completely, the user can browse the contact list and dial the contact number same as in local contacts.



Picture 111 - Downloading Cloud Phone book



Picture 112 - Browse contacts in cloud phone book

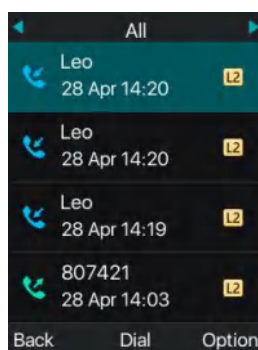
10.3 Call Log

The phone can store the call record (the quantity of storage varies according to different specifications). The user can press **[OK] >> [CallLog]** or press **[soft key-CallLog]** in standby mode to enter call log page, and the press **[Local Call Logs]** to open the call record and check the records of all incoming calls, outgoing calls and missed calls.

In the call logs interface, user may browse the call logs with **[Up/Down navigator keys]**.

Each call log record is presented with 'call type' and 'call party number / name'. User can check further call log detail by pressing **[OK]** key and dial the number with **[Dial]** key, or add the call log number to phonebook with pressing **[Option] >> [Add to Contact]** .

User can delete a call log by pressing **[Option] >> [Delete]** key and clear all call logs by pressing **[Delete All]** key.



Picture 113 - Call Log

Users can filter call logs by specific call types to narrow down search results. Use the **[Left/Right navigation keys]** to select and display one of the available call log categories.



- Missed Call Log



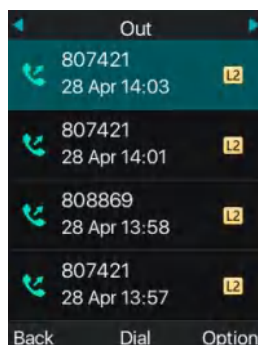
- Incoming Call Log



- Outgoing Call Log



- Forward Call Log

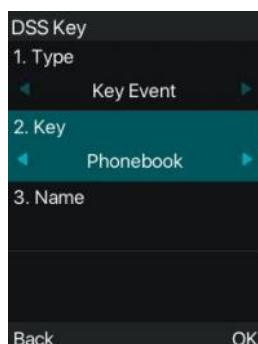


Picture 114 - Filter call record types

10.4 Function Key

Users can customize the side function keys of the handset. If a key is unconfigured, its settings can be modified by short pressing or long pressing. If the key has been assigned a function, long press it to edit the configuration.

Note: When the configuration key is configured as a function key type (such as PTT) triggered by a long press, and the health value is configured, users can press **[OK]** key in phone's standby mode to enter menu >> **[Basic]** >> **[Keyboard]** >> **[DSS Key Settings]** or configure and change the value of the function key on the web page.



Picture 115 - DSS LCD key Page Configuration Screen

The DSS Key could be configured as followings,

- Memory Key
 - Speed Dial / Intercom / Presence / Voice Mail / Call Park / BLF / Call Forward (to someone)
- BLF List
- XML Browser
- PTT
 - Intercom / Speed Dial / Multicast
- Key Event
 - VMail / DND / Hold / Xfer / Phonebook / Redial / Pickup / Join / Call Forward / Call Log / Flash / Memo / Headset / Release / Lock / SMS / Call Back / Hide DTMF / Intercom / Group Listening /

Prefix / Hot Desking / Agent / End / Disposition / Escalate / Trace / Handfree / Answer / Private Hold /
 Locla Contacts / LDAP Group / XML Group / Broadsoft Group / Record /Auto Headset / Acoustic
 Shield / Noise Cancellation / DSS Key Page1 / DSS Key Page2

- DTMF
- URL
- MCAST Paging

web page interface: **[Function key] >> [PTT Key]**.

Picture 116 - DSS settings

Moreover, user also can add the user-defined title for the DSS Keys, which is configured as Memory Key / Line / URL / Multicast / Prefix.

More detailed information *refers to* [12.25 Function Key](#) and [6.3 Appendix III –LED](#) .

10.5 Wi-Fi

The device has built-in Wi-Fi and supports wireless Internet access. For more details, please refer to [7.4 Connection Wi-Fi](#).

10.6 Headset

10.6.1 Wired Headset

The device supports wired headset with a Type-C headset jack, enabling functions such as playing incoming call alert tones through the headset and making calls via the headset. After the phone is connected to the headset, the status bar displays the headset icon  , indicating that the headset channel is available.

On the web page **[Phone settings] >> [Features] >> [Basic Settings]**, you can set the headset answering function, and the ring tone for headset.

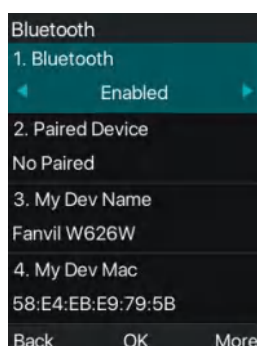
Note: W626W enables these two configurations by default.

Picture 117 - Headset function settings

10.6.2 Bluetooth Headset

The device has a built-in Bluetooth 5.0 module

- Press **[OK]** key to enter Menu >> **[Basic]** >> **[Bluetooth]** to enter the setup interface.
- Use the **[Left/Right navigation key]** to enable Bluetooth. Then use **[Up/Down navigation key]** to select “Paired Device”, if “No Paired” is displayed, press **[OK]** key/**[scan]** key to search and select the scanned device to connect.



Picture 118 - Bluetooth Settings Screen

The use of Bluetooth headset can be divided into three types: Answering call; Hanging up calls; Bluetooth redial.

- Answering calls

When the Bluetooth headset is connected to the phone, the incoming call can be answered by pressing the answer key on the Bluetooth headset.

- Hanging up calls

1) When talking with Bluetooth headset, the user can hang up the call by pressing the key on the Bluetooth headset.

2) When there is an incoming call, double-click the answer key to reject the call.

3) When the caller is in the ringing state, press the answer key of the headset to cancel the call.

- Bluetooth redial

When the Bluetooth headset is connected, double-click the answer key to redial the number dialed last time.

NOTICE! Some models may not support double - click redial function. Whether this function is supported or not, the user can check the instruction of the headset, or connect the Bluetooth headset to the phone, and double-click the answer key to see whether it redials or not.

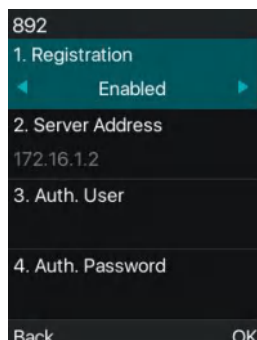
10.7 Advanced

10.7.1 Line Configurations

Users can configure the SIP lines on the phone.

When the phone is on the standby screen, press the **[OK]** key to enter menu >> **[Advanced]** >> enter the advanced settings password (default: “123”) >> **[Accounts]** >> select a SIP line >> **[Basic]** to access the

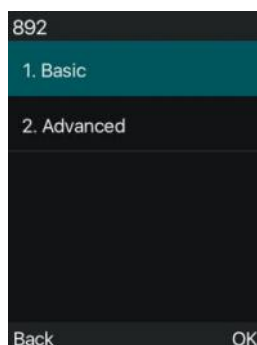
configuration page for the selected SIP line, as shown in the figure below. On this page, the user can modify the registration information of the SIP line. If the selected line is unregistered, configuring and saving the SIP registration parameters will cause the phone to send a registration request to the specified server and register the SIP account. Save the adjustment by pressing **[OK]** when done.



Picture 119 - SIP address and account information

Up to now, all basic configurations have been completed. The phone will be ready for normal use after saving the settings.

Users who want to configure more options should use web management portal to modify or system in accounts on the individual line to configure those options.



Picture 120 - Configure Advanced Line Options

10.7.2 Network Settings

When the phone is in standby mode, press the **[OK]** key to enter menu >> **[Advanced]** >> enter the advanced settings password (default password: "123") >> **[Network]** to access the advanced network settings page of the phone.

10.7.2.1 Network Settings

On the advanced settings page, choose and press **[Network]**, and then choose **[Network Settings]** to modify the basic network configurations.

■ IP Mode

There are 3 network protocol mode options, IPv4, IPv6 and IPv4 & IPv6.

Users can set it on the web page **[Network]** >> **[Basic]**. Select Wi-Fi for the net type, and then the user can set the network mode.

Network Mode ?

Network Mode: IPv4 Only ▾

IPv4 Network Status

IP:

Subnet mask: 255.255.254.0

Picture 121 - Network mode Settings

■ IPv4

In IPv4 mode, there are 2 connection mode options: DHCP and Static IP.

IPv4 Settings

Static IP ☐ DHCP ☒

DHCP Hostname:

Enable Vendor Identifier: Disabled ▾

Vendor Identifier:

DNS Server Configured by: DHCP ▾

Primary DNS Server:

Secondary DNS Server:

DNS Domain:

Picture 122 - DHCP network mode

When using DHCP mode, the phone will get the IP address from DHCP server (router).

- Use DHCP DNS: It is enabled as default. “Enable” means phone will get DNS address from DHCP server and “disable” means not.
- Use DHCP time: It is disabled as default. “Enable” to manage the time of get DNS address from DHCP server and “disable” means not.

IPv4 Settings

Static IP ☒ DHCP ☐

IP:

Subnet mask:

Default gateway:

Primary DNS Server:

Secondary DNS Server:

DNS Domain:

Picture 123 - Static IP network mode

When using Static IP mode, user must configure the IP address manually.

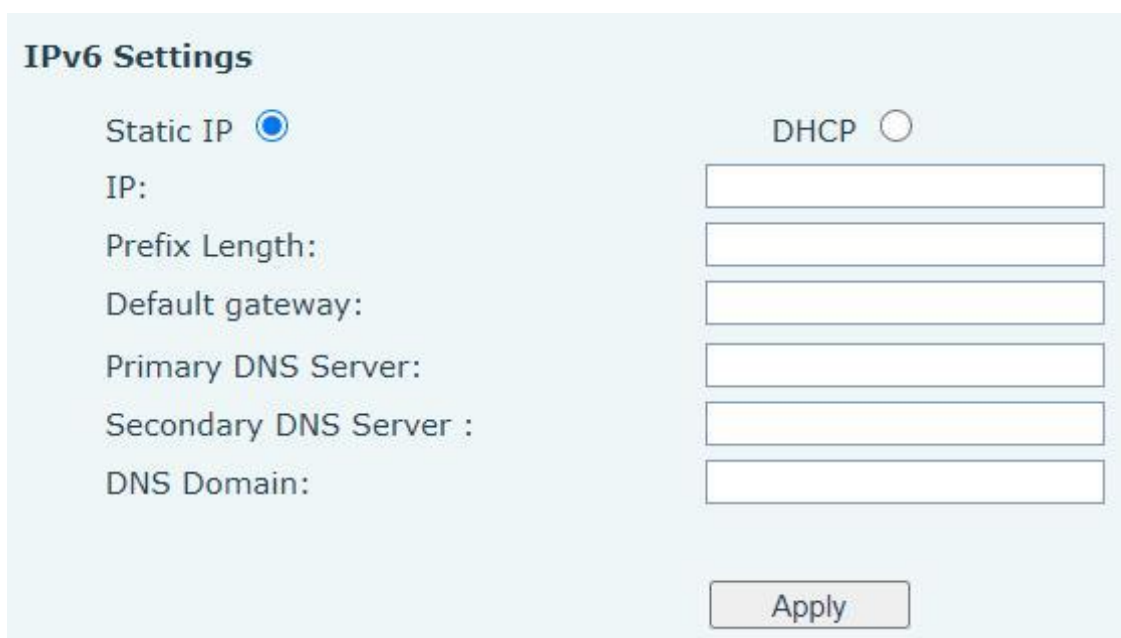
- IP Address: Phone IP address.

- Mask: sub mask of your LAN.
- Gateway: The gateway IP address. Phone could access the other network via it.
- Primary DNS: Primary DNS address. The default is 8.8.8.8, Google DNS server address.
- Secondary DNS: When primary DNS is not available, Secondary DNS will work.

■ IPv6

In IPv6, there are 2 connection mode options, DHCP and Static IP.

- DHCP configuration refers to IPv4 introduction in last page.
- Static IP configuration is almost same as the Static IP configuration of IPv4, except the IPv6 Prefix.
- IPv6 Prefix: IPv6 prefix, it is similar with mask of IPv4.



IPv6 Settings

Static IP ☒ DHCP ☐

IP:

Prefix Length:

Default gateway:

Primary DNS Server:

Secondary DNS Server :

DNS Domain:

Apply

Picture 124 - IPv6 Static IP network mode

10.7.2.2 VPN

Virtual Private Network (VPN) is a technology to allow device to create a tunneling connection to a server and becomes part of the server's network. The network transmission of the device may be routed through the VPN server.

For some users, especially enterprise users, a VPN connection might be required to be established before activate a line registration. The device supports two VPN modes, Layer 2 Transportation Protocol (L2TP) and OpenVPN.

On the **[Advanced]** > **[Network]** page, select and press **[Virtual Private Network]** to configure VPN-related settings. The VPN connection must be configured and started (or stopped) from the device web portal.

■ L2TP

NOTICE! The device only supports non-encrypted basic authentication and non-encrypted data tunneling. For users who need data encryption, please use OpenVPN instead.

To establish a L2TP connection, users should log in to the device web portal, open web page **[Network] >> [VPN]**. In VPN Mode, check the **[Enable VPN]** option and select **[L2TP]**, then fill in the L2TP server address, Authentication Username, and Authentication Password in the L2TP section. Press **[Apply]** and the device will try to connect to the L2TP server.

When the VPN connection established, the VPN IP Address should be displayed in the VPN status. There may be the delay of the connection establishment. User may need to refresh the page to update the status.

Once the VPN is configured, the device will try to connect with the VPN automatically when the device boots up every time until user disable it. Sometimes, if the VPN connection does not establish immediately, user may try to reboot the device and check if VPN connection established after reboot.

■ OpenVPN

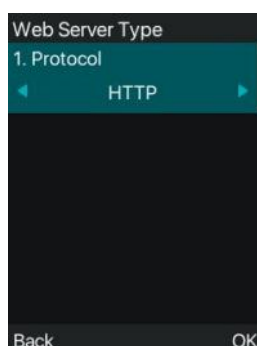
To establish an OpenVPN connection, user should get the following authentication and configuration files from the OpenVPN hosting provider and name them as the following,

OpenVPN Configuration file: // client.ovpn
CA Root Certification: // ca.crt
Client Certification: // client.crt
Client Key: // client.key

User then upload these files to the device in the web page **[Network] >> [VPN]**, select OpenVPN Files. Then user should check “Enable VPN” and select **[OpenVPN]** in VPN Mode and click **[Apply]** to enable OpenVPN connection. Same as L2TP connection, the connection will be established every time when system rebooted until user disable it manually.

10.7.2.3 Web Server Type

On the **[Advanced] >> [Network]** page, select and press **[Web Server Type]** to modify the protocol, as shown in the figure below. Configure the Web Server mode to be HTTP or HTTPS and will be activated after the reboot. Then user could use HTTP / HTTPS protocol to access phone web page.



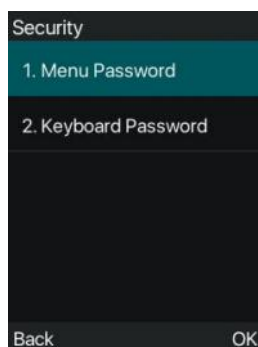
Picture 125 - The phone configures the web server type

10.7.3 Set The Secret Key

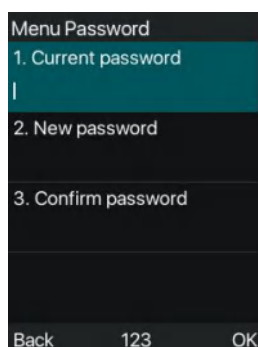
When the device is in the default standby mode,

- Press **[OK]** key to enter menu >> **[Advanced]**.
- As default, the Advance setting password is 123. Select **[Security]**, then user will see the follow page

“Security”.



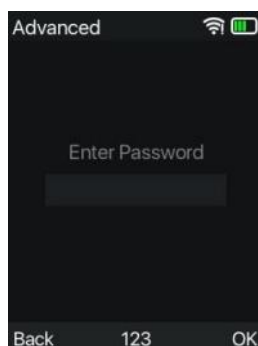
Picture 126 - Menu password



Picture 127 - Set menu password

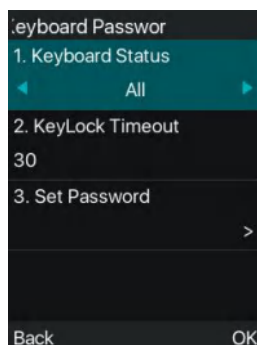
Menu password is the permission for accessing the advanced setting.

- **[Current password]** is the password user configured before. If no configuration before, the default password is 123.
- **[New password]** is the new password user to use.
- After configuring the menu password, it will work immediately.



Picture 128 - Keyboard lock password

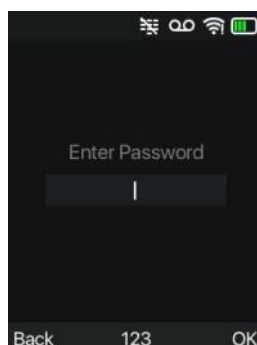
The keyboard password is used to unlock the phone keyboard once it's locked.



Picture 129 - Set keyboard lock password

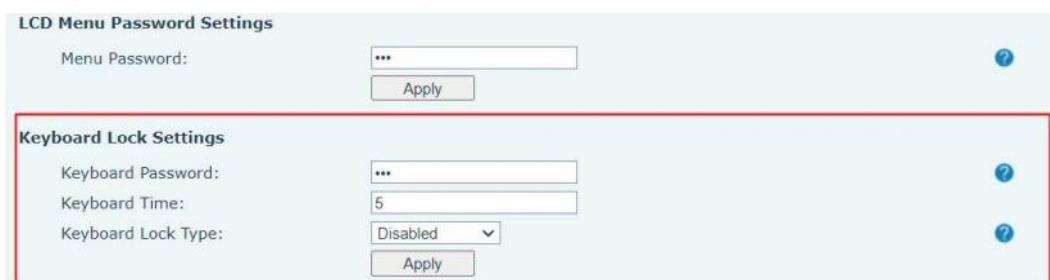
In the phone LCD screen, user could only set the keyboard password to enable or disable.

- Enter **[Keyboard password]** setting by pressing **[OK]** key after password entered. If no menu password configuration before, it is 123 as default.
- If the menu password is correct, phone will go to keyboard password interface. As default, the keyboard password is disabled. When it is enabled, the keyboard will be locked after timeout.
- If user does not configure the keyboard lock time, (it is 0 as default). Long pressing **[#]** will lock the phone. There will be a lock icon  in the top of LCD. Phone will reminder “Enter Password” after pressing any keys.



Picture 130 - Phone keypad lock password input interface

Meanwhile, users can also configure the phone password on the web interface. Log in to the phone web page >> **[Phone Settings]** >> **[Advanced]** to enter the advanced settings page. Scroll down to find the options for setting the menu password and keypad password.



Picture 131 - Web keyboard lock password settings

10.7.4 Auto provision

Phone Web page: Login and go to **[System] >> [Auto provision]**.

Basic Settings

CPE Serial Number: 00100400FV02001000000c383e904a7f

Authentication Name:

Authentication Password:

Configuration File Encryption Key:

General Configuration File Encryption Key:

Download Fail Check Times: 1

Update Contact Interval: 720 (0, >=5) Minute

Save Auto Provision Information: ☐

Download CommonConfig enabled: ☒

Enable Server Digest: ☐

Display Provision Prompt: Disable All Provision Prompt

Provision Config Priority: Normal

DHCP Option >>

DHCPv6 Option >>

SIP Plug and Play (PnP) >>

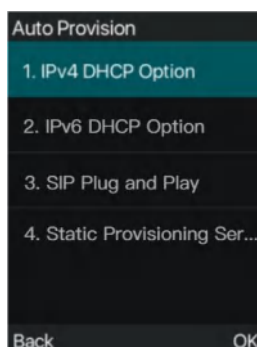
Static Provisioning Server >>

Autoprovision Now >>

TR069 >>

Picture 132 - Page auto provision Settings

Phone LCD: Press **[OK]** key in standby mode to enter menu >> **[Advanced] >> [Maintenance] >> [Auto Provision]**.



Picture 133 - Phone auto provision settings

Fanvil devices support SIP Plug and Play, DHCP options, Static provision, TR069. If all of the 4 methods are enabled, the device will be upgraded according to the method obtained first.

Transferring protocol: FTP, TFTP, HTTP, HTTPS

This article only briefly introduces automatic deployment. For details, please refer to the document **Fanvil Auto Provision**.

Table 12 - Auto Provision

Parameters	Description
Basic settings	
CPE Serial Number	Display the device SN
Authentication Name	The user name of provision server

Authentication Password	The password of provision server
Configuration File Encryption Key	If the device configuration file is encrypted , user should add the encryption key here.
General Configuration File Encryption Key	If the common configuration file is encrypted, user should add the encryption key here.
Download Fail Check Times	If there download is failed, phone will retry with the configured times.
Update Contact Interval	Phone will update the phonebook with the configured interval time. If it is 0, the feature is disabled.
Save Auto Provision Information	Save the HTTP/HTTPS/FTP user name and password. If the provision URL is kept, the information will be kept.
Download Common Config enabled	Whether phone will download the common configuration file.
Enable Server Digest	When the feature is enable, if the configuration of server is changed, phone will download and update.
Display Provision Prompt	It configures the Display Provision Prompt
Provision Config Priority	Provision Config Priority
DHCP Option	
Option Value	Configure DHCP option, DHCP option supports DHCP custom option DHCP option 66 DHCP option 43, 3 methods to get the provision URL. The default is Option 66.
Custom Option Value	Custom Option value is allowed from 128 to 254. The option value must be same as server define.
Enable DHCP Option 120	Use Option120 to get the SIP server address from DHCP server.
DHCPv6 Option	
Option Value	DHCP OPTION type for Auto Provisioning.
Custom Option Value	
SIP Plug and Play (PnP)	
Enable SIP PnP	Whether enable PnP or not. If PnP is enable, phone will send a SIP SUBSCRIBE message with broadcast method. Any server can support the feature will respond and send a SIP Notify with URL to phone. Phone could get the configuration file with the URL.
Server Address	Broadcast address. As default, it is 224.0.0.0.
Server Port	PnP port
Transport Protocol	PnP protocol, TCP or UDP.
Update Interval	PnP message interval
Static Provisioning Server	
Server Address	Provisioning server address. Support both IP address and domain address.

Configuration File Name	The configuration file name. If it is empty, phone will request the common file and device file which is named as its MAC address. The file name could be a common name, \$mac, \$input. The file format supports CFG/TXT/XML.
Protocol Type	Transferring protocol type , supports FTP、TFTP、HTTP and HTTPS
Update Interval	Configuration file update interval time. As default it is 1, means phone will check the update every 1 hour.
Update Mode	Provision Mode. 1. Disabled. 2. Update after reboot. 3. Update after interval.
TR069	
Enable TR069	Enable TR069 after selection
ACS Server Type	There are 2 options Serve type, Common and CTC.
ACS Server URL	ACS server address
ACS User	ACS server username (up to is 59 characters)
ACS Password	ACS server password (up to is 59 characters)
Enable TR069 Warning Tone	If TR069 is enabled, there will be a prompt tone when connecting.
TLS Version	TLS version (TLS 1.2, TLS 1.3)
INFORM Sending Period	INFORM signal interval time. It ranges from 1s to 999999s
STUN Server Address	Configure STUN server address
STUN Enable	To enable STUN server for TR069

10.7.5 Firmware Upgrade

- Web page: Login phone web page, go to **[System] >> [Upgrade]**

Software upgrade

Current Software Version: 0.0.12

System Image File:

Upgrade Server

Enable Auto Upgrade: ☒

Upgrade Server Address1:

Upgrade Server Address2:

Update Interval: Hour(s)

Firmware Information

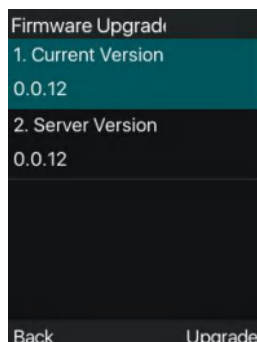
Current Software Version: 0.0.12

Server Firmware Version: 0.0.12

New Firmware Information:

Picture 134 - Web page firmware upgrade

- LCD interface: press **[OK]** in phone's standby mode to enter menu>> **[Advanced]** >> **[Firmware Upgrade]**.



Picture 135 - Firmware upgrade information display

Table 13 - Firmware upgrade

Parameter	Description
Upgrade server	
Enable Auto Upgrade	Enable automatic upgrade, If there is a new version TXT and new software firmware on the server, phone will show a prompt upgrade message after Update Interval.
Upgrade Server Address1	Set available upgrade server address.
Upgrade Server Address2	Set back up available upgrade server address.
Update Interval	Set Update Interval. Enable Auto Upgrade and configure the Update Interval. If the server has a new firmware, the phone will prompt for upgrade at the interval.
Firmware Information	
Current Software Version	It will show Current Software Version.
Server Firmware Version	It will show Server Firmware Version.
[Upgrade] key	If there is a new version TXT and new software firmware on the server, the page will display version information and upgrade key will become available; Click [Upgrade] key to upgrade the new firmware.
New version description information	When there is a corresponding TXT file and version on the server side, the TXT and version information will be displayed under the new version description information.

- The file requested from the server is a TXT file called **[vendor]_[model]_hwv1_0.txt**. hwv followed by the hardware version number, it will be written as hw1_0 if no difference on hardware. All Spaces in the filename are replaced by underline.
For example, the TXT file name requested by W626W phone is fanvil_w626w_hwv1_1.txt.
- The URL requested by the phone is HTTP:// server address/, the new version and the requested file should be placed in the download directory of the HTTP server, as shown in the figure:

 fanvil_w626w_hwv1_1.txt	2026/4/21 12:26
 w62xw-linkvil-release-9011-a00e-0.0.12-krvT2026-04-20-00.43.17.z	2026/4/21 12:26

Picture 136 - HTTP server download directory

- TXT file format must be UTF-8
- The format of the file “ vendor_model_hwv1_0.txt ” is as follows:
Version=2.16.10 #Firmware version
Firmware=xxx/xxx.z #URL, Relative paths are supported and absolute paths are possible,
distinguished by the presence of protocol headers.
BuildTime=2022.05.06 20:00
Info=TXT

Xxxxx

Xxxxx

Xxxxx

Xxxxx

- After the update cycle interval time arrives, if the server has available files and versions, the phone will prompt as shown below, click View to view the version information and upgrade.

10.7.6 Factory Reset

1) Method 1: The phone is in default standby mode,

- Press **[OK]** key to enter menu >> **[Advanced]** >> enter the menu password(it is defaulted set as"123").
- Choose and press **[Factory Reset]**, and select the file to be cleared.
 - Press **[OK]** to clear after completion. When you select clear configuration file and clear all, the phone will restart automatically after clearing.

2) Method 2: In standby, press and hold the **[OK]** key for 6 seconds to perform the reset operation.

11 Web Configurations

11.1 Web Page Authentication

The user can log into the web page of the phone to manage the user's phone information and operate the phone. Users must provide the correct user name and password to log in. For detailed login procedures to access the phone web page, refer to [7.9 Web Management](#).

11.2 System >> Information

User can get the system information of the device in this page including,

- Model
- Hardware Version
- Software Version
- Uboot
- Uptime
- Memory information
- System time

And contains network status,

- Network Mode
- Wi-Fi MAC
- Bluetooth MAC
- Wi-Fi IP
- Subnet Mask
- Default Gateway

Besides, contains SIP account status,

- SIP User
- SIP account status (Registered / Inactive / Trying / Timeout)

11.3 System >> Account

On this page the user can change the password for the login page.

Users with administrator rights can also add or delete users, manage users, and set permissions and passwords for new users.

11.4 System >> Configurations

On this page, users with administrator privileges can view, export, or import the phone configuration, or restore the phone to factory Settings.

- **Export Configuration**

Follow the on-screen instructions. Left-click at the corresponding position to view the configuration file content, and right-click to save the configuration file.

■ Import Configuration

Click **[Select]** here and choose a configuration file. The settings in the new configuration file will overwrite the current ones. Click **[Import]** to complete the process. The new configuration takes effect immediately after successful import.

■ Clear Configurations

Select the module in the configuration file to clear.

SIP: account configuration.

Basic Network: Basic network configuration.

AutoProvision: automatically upgrades the configuration

TR069:TR069 related configuration

MMI: MMI module, including authentication user information, web access protocol, etc.

Dsskey: DSS Key configuration

■ Clear Userdata

Select the local data table to be cleared, all selected by default.

■ Clear ETC

Select the ETC files to be cleared, including authentication cert, https, SIP Accounts, OpenVPN, mmiset, language, and wallpaper.

■ Reset Phone

The phone data will be cleared, including configuration and database tables.

11.5 System >> Upgrade

Upgrade the phone software version, customized ringtone, background, boot logo icon, etc, can also be upgraded to delete the file. Ring tone support “.wav” format.

11.6 System >> Auto Provision

The Auto Provision settings help IT manager or service provider to easily deploy and manage the devices in mass volume. For the detail of Auto Provision, please refer to [10.7.4 Auto provision](#).

11.7 System >> Tools

Tools provided in this page help users to identify issues at trouble shooting. Please refer to [13 Trouble Shooting](#) for more detail.

11.8 System >> Reboot Phone

This page can reboot the phone.

12 Network >> Basic

This page allows users to configure network connection types and parameters.

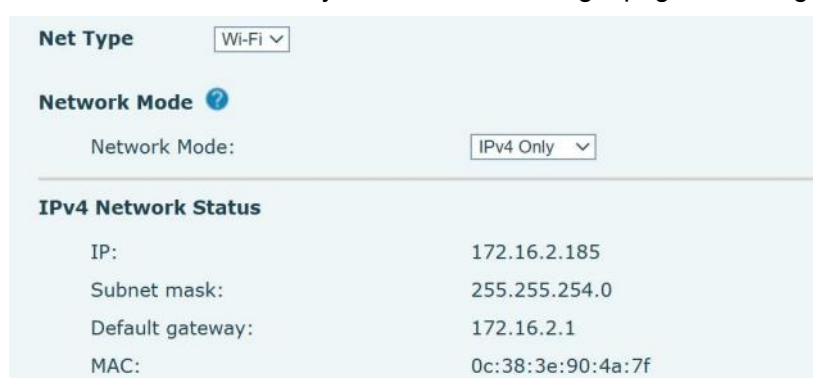
12.1 Network >> Wi-Fi Settings

The default network priority is Wi-Fi.

The current device supports coexistence of Wi-Fi and Ethernet, and users can log in to the web page with any network address for configuration.

For example, Wi-Fi access IP is 172.16.3.138 and Ethernet access IP is 172.16.7.116.

Page login 172.16.7.116, 172.16.3.138 Any network address login page for configuration.



Net Type Wi-Fi

Network Mode ?

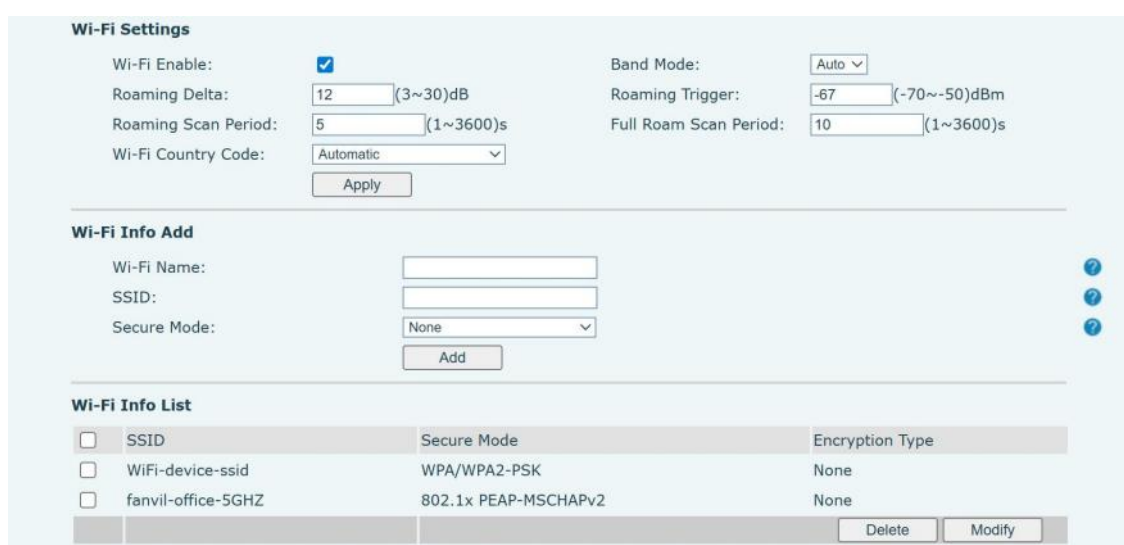
Network Mode: IPv4 Only

IPv4 Network Status

IP:	172.16.2.185
Subnet mask:	255.255.254.0
Default gateway:	172.16.2.1
MAC:	0c:38:3e:90:4a:7f

Picture 137 - Network Priority

[Network] >> [Wi-Fi Settings] page can turn on Wi-Fi, add Wi-Fi information, and view the wireless network list.



Wi-Fi Settings

Wi-Fi Enable: ☒

Roaming Delta: (3~30)dB

Roaming Scan Period: (1~3600)s

Wi-Fi Country Code: Automatic

Band Mode: Auto

Roaming Trigger: (-70~-50)dBm

Full Roam Scan Period: (1~3600)s

Apply

Wi-Fi Info Add

Wi-Fi Name:

SSID:

Secure Mode: None

Add

Wi-Fi Info List

☐	SSID	Secure Mode	Encryption Type
☐	WiFi-device-ssid	WPA/WPA2-PSK	None
☐	fanvil-office-5GHZ	802.1x PEAP-MSCHAPv2	None

Delete Modify

Picture 138 - Wi-Fi Settings

12.2 Network >> Service Port

This page provides settings for Web page login protocol, protocol port settings and RTP port.

Picture 139 - Service Port Settings

Table 14 - Service port

Parameter	Description
Web Server Type	Reboot to take effect after settings. Optionally, the web page login is HTTP/HTTPS.
Web Logon Timeout	Default as 15 minutes, the timeout will automatically exit the login page, need to login again.
Web auto login	After the timeout does not need to enter a user name password, will automatically login to the web page.
HTTP Port	The default is 80. If you want system security, you can set ports other than 80. Such as :8080, web page login: HTTP://ip:8080
HTTPS Port	The default is 443, the same as the HTTP port.
TLS Version	The version of TLS.
RTP Port Range Start	The value range is 1025 to 65535. The value of RTP port starts from the initial value set. For each call, the value of voice and video port is added 2.
RTP Port Quantity	Number of calls.

12.3 Network >> VPN

Users can configure a VPN connection on this page. See [10.7.2.2 VPN](#) for more details.

12.4 Network >> Advanced

Advanced network Settings are typically configured by the IT administrator to improve the quality of the phone service. For configuration, query the [10.7 Advanced](#) Settings.

12.5 Line >> SIP

Configure the Line service configuration on this page.

Table 15 - Line configuration on the web page

Parameters	Description
Register Settings	
Line Status	Display the current line status at page loading. To get the up to date line status, user has to refresh the page manually.
Activate	Whether the service of the line is activated.
Username	Enter the username of the service account.
Authentication User	Enter the authentication user of the service account.
Display Name	Enter the display name to be sent in a call request.
Authentication Password	Enter the authentication password of the service account.
Realm	Enter the SIP domain if requested by the service provider.
Server Name	Input server name.
SIP Server 1	
Server Address	Enter the IP or FQDN address of the SIP server
Server Port	Enter the SIP server port, default is 5060
Transport Protocol	Set up the SIP transport line using TCP or UDP or TLS.
Registration Expiration	Set SIP expiration date.
Proxy Server Address	Enter the IP or FQDN address of the SIP proxy server.
Proxy Server Port	Enter the SIP proxy server port, default is 5060.
Proxy User	Enter the SIP proxy user.
Proxy Password	Enter the SIP proxy password.
SIP Server 2	
Server Address	Enter the IP or FQDN address of the SIP server.
Server Port	Enter the SIP server port, default is 5060.
Transport Protocol	Set up the SIP transport line using TCP or UDP or TLS.
Registration Expiration	Set SIP expiration date.
Backup Proxy Server Address	Enter the IP or FQDN address of the backup proxy server.
Backup Proxy Server Port	Enter the backup proxy server port, default is 5060.
Basic Settings	
Enable Auto Answering	Enable auto-answering, the incoming calls will be answered automatically after the delay time.
Auto Answering Delay	Set the delay for incoming call before the system automatically answered it.
Call Forward	Enable unconditional call forward, all incoming calls will be forwarded to the

Unconditional	number specified in the next field.
Call Forward Number for Unconditional	Set the number of unconditional call forward.
Call Forward on Busy	Enable call forward on busy, when the phone is busy, any incoming call will be forwarded to the number specified in the next field.
Call Forward Number for Busy	Set the number of call forward on busy .
Call Forward on No Answer	Enable call forward on no answer, when an incoming call is not answered within the configured delay time, the call will be forwarded to the number specified in the next field.
Call Forward Number for No Answer	Set the number of call forward on no answer.
Call Forward Delay for No Answer	Set the delay time of not answered call before being forwarded.
Transfer Timeout	Set the timeout of call transfer process.
Conference Type	Set the type of call conference, Local=set up call conference by the device itself, maximum supports two remote parties, Server=set up call conference by dialing to a conference room on the server.
Server Conference Number	Set the conference room number when conference type is set to be Server.
Subscribe For Voice Message	Enable the device to subscribe a voice message waiting notification, if enabled, the device will receive notification from the server if there is voice message waiting on the server.
Voice Message Number	Set the number for retrieving voice message.
Voice Message Subscribe Period	Set the interval of voice message notification subscription.
Enable Hotline	Enabling hotline configuration, the device will dial to the specific number immediately at audio channel opened by off-hook handset or turn on hands-free speaker or headset.
Hotline Delay	Set the delay for hotline before the system automatically dialed it.
Hotline Number	Set the hotline dialing number.
Dial Without Registered	Set call out by proxy without registration.
Enable Missed Call Log	If enabled, the phone will save missed calls into the call history record.
DTMF Type	Set the DTMF type to be used for the line.
DTMF SIP INFO Mode	Set the SIP INFO mode to send '*' and '#' or '10' and '11'.
Request With Port	Whether the URI carries port number.
Enable DND	Enable Do-not-disturb, any incoming call to this line will be rejected automatically.

Use VPN	Set the line to use VPN restrict route.
Use STUN	Set the line to use STUN for NAT traversal.
Enable Failback	Whether to switch to the primary server when it is available.
Failback Interval	A Register message is used to periodically detect the time interval for the availability of the main Proxy.
Signal Failback	Multiple proxy cases, whether to allow the invite/register request to also execute failback.
Signal Retry Counts	The number of attempts that the SIP Request considers proxy unavailable under multiple proxy scenarios.
Codecs Settings	Set the priority and availability of the codecs by adding or remove them from the list.
Video Codecs	Set the priority and availability of the codec by adding or removing in the list.
Advanced Settings	
Use Feature Code	When this setting is enabled, the features in this section will not be handled by the device itself but by the server instead. In order to control the enabling of the features, the device will send feature code to the server by dialing the number specified in each feature code field.
Enable DND	It configures the DND on code to activate the server-side DND feature.
DND Disable	It configures the DND off code to deactivate the server-side DND feature.
Enable Call Forward Unconditional	It configures the always forward on code to activate the server-side always forward feature.
Disable Call Forward Unconditional	It configures the always forward off code to deactivate the server-side always forward feature.
Enable Call Forward on Busy	It configures the busy forward on code to activate the server-side busy forward feature.
Disable Call Forward on Busy	It configures the busy forward off code to deactivate the server-side busy forward feature.
Enable Call Forward on No Answer	It configures the no answer forward on code to activate the server-side no answer forward feature.
Disable Call Forward on No Answer	It configures the no answer forward off code to deactivate the server-side no answer forward feature.
Enable Blocking Anonymous Call	It configures the Blocking Anonymous Call On code activate the server-side Blocking Anonymous Call feature.
Disable Blocking Anonymous Call	It configures the Blocking Anonymous Call Off code deactivate the server-side Blocking Anonymous Call feature.
Call Waiting On Code	It configures the call waiting on code to activate the server-side call waiting feature.

Call Waiting Off Code	It configures the CallWaiting Off code to deactivate the server-side call waiting feature.
Send Anonymous On Code	It configures the Anonymous Call On code activate the server-side Anonymous Call feature.
Send Anonymous Off Code	It configures the Anonymous Call Off code deactivate the server-side Anonymous Call feature.
Enable Session Timer	When the call timer is enabled, when the configuration item is switched on, the phone periodically sends the message and terminates the call without a reply.
Session Timeout	Set the session timer timeout period.
Enable BLF List	Open the BLF List function.
BLF List Number	Set the BLF List number and bind the BLF List of this number to the DSS Key.
Response Single Codec	If setting enabled, the device will use single codec in response to an incoming call request.
BLF Server	The BLF server is used in conjunction with the BLF list.
Keep Alive Type	Set the line to use dummy UDP or SIP OPTION packet to keep NAT pinhole opened.
Keep Alive Interval	Set the keep alive packet transmitting interval.
Keep Authentication	Keep the authentication parameters from previous authentication.
Blocking Anonymous Call	Reject any incoming call without presenting caller ID.
RTP Encryption(SRTP)	RTP encryption. When this configuration is open, the voice of the phone is encrypted, the conversation is normal, and the phone is not heard by the grab bag.
Enable OSRTP	When RTP Encryption(SRTP) is Optional, It decides whether SRTP applies OSRTP or Optional.It is OSRTP when it is enabled, otherwise it is Optional.
Proxy Require	It configures the Proxy-Require header of sip request. Valid Value: Within 63 characters.
User Agent	Set the user agent, the default is Model with Software Version.
Specific Server Type	Set the line to collaborate with specific server type.
SIP Version	Set the SIP version.
Anonymous Call Standard	Set the standard to be used for anonymous.
Local Port	Set the local port.
Ring Type	Set the ring tone type for the line.
Enable user=phone	Sets user=phone in SIP messages.
Use Tel Call	Set use tel call.
Auto TCP	Using TCP protocol to guarantee usability of transport for SIP messages above 1500 bytes.

Enable Rport	Set the line to add rport in SIP headers.
Enable PRACK	Set the line to support PRACK SIP message.
Call-ID Format	It configures the Call-ID Header format of SIP message. Valid input: \$id、\$id@\$ip.
DNS Mode	Select DNS mode, A, SRV, NAPTR.
Enable Long Contact	Allow more parameters in contact field per RFC 3840.
Enable Strict Proxy	Enables the use of strict routing. When the phone receives packets from the server, it will use the source IP address, not the address in via field.
Convert URI	Enable URI conversion Whether or not.
Use Quote in Display Name	Whether to add quote in display name, i.e. "Fanvil" vs Fanvil.
Enable GRUU	Support Globally Routable User-Agent URI (GRUU).
Sync Clock Time	Time Sync with server.
Enable Inactive Hold	With the post-call hold capture package enabled, you can see that in the INVITE package, SDP is inactive.
Caller ID Header	Set the Caller ID Header.
Use 182 Response for Call waiting	Set the device to use 182 response code at call waiting response.
Enable Feature Sync	Feature Sync with server.
Enable SCA	Enable/Disable SCA (Shared Call Appearance).
TLS Version	Choose TLS Version.
uaCSTA Number	Set uaCSTA Number.
Enable Preview	Whether to enable preview.
Preview Mode	Set the preview mode.
Enable Click To Talk	With the use of special server, click to call out directly after enabling.
Enable ChangePort	Whether to enable changeport.
VQ Name	Set VQ name.
VQ Server	Set VQ server.
VQ Server Port	Set VQ server port..
VQ Http/Https server	Set VQ http/https server.
Server Expire	Use the timeout of the server.
PickUp Number	Set the scramble number when the Pickup is enabled.
JoinCall Number	Set JoinCall Number.
Intercom Number	Set Intercom Number.
CallPark Number	CallPark number. Keep the call to the configured number, record the number of voice broadcasts, and then use the number of other terminal call records to retrieve the call.

Retrieve Number	Set retrieve number.
Unregister On Boot	Whether to enable logout function.
Enable MAC Header	When opening the registration, are IP package and user agent with MAC.
Enable Register MAC Header	When opening the registration, is user agent with MAC.
BLF Dialog Strict Match	Whether to enable accurate matching of BLF sessions.
PTime(ms)	Set whether to bring ptime field, default no.
Enable Deal 180	Set whether to enable deal 180.
Transaction Timer T1	It configures the SIP Transaction Timer T1(in millionseconds).
Transaction Timer T2	It configures the SIP Transaction Timer T2(in millionseconds).
Transaction Timer T4	It configures the SIP Transaction Timer T4(in millionseconds).
SIP Global Settings	
Strict Branch	Set up to strictly match the Branch field.
Enable Group	Set open group.
Enable RFC4475	Set to enable RFC4475.
Enable Strict UA Match	Enable strict UA matching.
Registration Failure Retry Time	Set the registration failure retry time.
Local SIP Port	Modify the phone SIP port.
Enable uaCSTA	Enable uaCSTA.
SMS direct Enabled	Set enable or disable or show only a period of time.
SMS Save Enabled	Whether to enable SMS save.
SMS Ring Enabled	Whether to enable SMS ring.
SMS Display Time	Set SMS display time.

12.6 Line >> SIP Hotspot

Please refer to [9.9 SIP Hotspot](#)

12.7 Line >> Dial Plan

Basic Settings

☒ Press # to invoke dialing
 ☐ Dial Fixed Length 11 to Send
 ☒ Send after 10 second(s)(3~30)
 ☐ Press # to Do Blind Transfer
 ☐ Blind Transfer on Onhook
 ☐ Attended Transfer on Onhook
 ☐ Attended Transfer on Conference Onhook
 ☐ Enable E.164

Apply

Picture 140 - Dial plan settings

Table 16 - Phone 5 dialing methods

Parameters	Description
Press # to invoke dialing	The user dials the other party's number and then adds the # number to dial out.
Press # to Do Blind Transfer	With "#" end blind transfer, enter the number in the transfer dial, then enter # for the blind transfer.
Dial Fixed Length	The number entered by the user is automatically dialed out when it reaches a fixed length.
Timeout dial	The system dials automatically after timeout.
Enable E.164	Please refer to E. 164 standard specification.

Add dialing rules:

Picture 141 - Custom setting of dial - up rules

Table 17 - Dial - up rule configuration table

Parameters	Description
Dial rule	There are two types of matching: Full Matching or Prefix Matching. In Full matching, the entire phone number is entered and then mapped per the Dial Peer rules. In prefix matching, only part of the number is entered followed by T. The mapping with then take place whenever these digits are dialed. Prefix mode supports a maximum of 30 digits.
Note: Two different special characters are used. ■ x -- Matches any single digit that is dialed. ■ [] -- Specifies a range of numbers to be matched. It may be a range, a list of ranges separated by commas, or a list of digits.	
Destination	Set Destination address. This is for IP direct.
Port	Set the Signal port, and the default is 5060 for SIP.
Alias	Set the Alias. This is the text to be added, replaced or deleted. It is an optional item.
Note: There are four types of aliases. ■ all: xxx – xxx will replace the phone number. ■ add: xxx – xxx will be dialed before any phone number. ■ del: The characters will be deleted from the phone number. ■ rep: xxx – xxx will be substituted for the specified characters.	
Suffix	Characters to be added at the end of the phone number. It is an optional item.
Length	Set the number of characters to be deleted. For example, if this is set to 3, the phone will delete the first 3 digits of the phone number. It is an optional item.

For Example

This feature allows the user to create rules to make dialing easier. There are several different options for dial rules. The examples below will show how this can be used.

Example 1: All Substitution -- Assume that it can make a direct IP call to IP address 172.168.2.208. Using this feature, 123 can be substituted for 172.168.2.208.

User-defined Dial Plan Table							
Index	Digit Map	Call	Match to Send	Line	Alias Type: Number(length)	Suffix	Media
1	"123"	Out	No	SIP DIALPEER(172.16.1.97:7060)			Default

Picture 142 - Dial rules table (1)

Example 2: Partial Substitution -- To dial a long distance call to Beijing requires dialing area code 010 before the local phone number. Using this feature 1 can be substituted for 010. For example, to call 62213123 would only require dialing 162213123 instead of 01062213123.

User-defined Dial Plan Table							
Index	Digit Map	Call	Match to Send	Line	Alias Type: Number(length)	Suffix	Media
1	"1T"	Out	No	Fanvil@SIP1	rep:010(1)		Default

Picture 143 - Dial rules table (2)

Example 3: Addition -- Two examples are shown. In the first case, it is assumed that 0 must be dialed before any 11 digit numbers beginning with 131. In the second case, it is assumed that 0 must be dialed before any 11 digit numbers beginning with 135, 136, 137, 138, or 139.

Two different special characters are used:

x -- Matches any single digit that is dialed.

[] -- Specifies a range of numbers to be matched. It may be a range, a list of ranges separated by commas, or a list of digits.

12.8 Line >> Action Plan

Action Plan application: a technical implementation defined and designed by Fanvil for remote control and behavior linkage between Fanvil terminal equipment and other equipment. That is, when an event occurs on the Fanvil terminal, the terminal can perform an action, and this action is completed according to a Plan rule.

Log in to the phone web page, access **[Line] >> [Action Plan]**, and configure the linkage plan rules.

Action Plan Add

Action:

Default

?

Number:

Default

?

Direction:

Video

?

MCAST Codec:

MCAST-Xfer

?

Username:

?

URL:

?

Type:

Early

?

Line:

AUTO

?

Password:

?

UserAgent:

?

Add

Action Plan Option

▼

Delete

Modify

User-defined Action Plan Table

Index	Action	Number	Type	Direction	Line	URL Index	Username	URL	UserAgent
-------	--------	--------	------	-----------	------	-----------	----------	-----	-----------

Picture 144 - Action Plan

Table 18 - Action Plan

Parameter	Description
Action	Default: when the rule is triggered, the phone displays video or converts multicast according to the RTSP URL or multicast address port set by the website. Video: when the rule is triggered, the phone accesses the RTSP URL configured by the URL to display the video. MCAST-XFER: when the rule is triggered, the phone converts the incoming call or multicast into multicast and sends it to the set multicast address port. Record: the phone automatically turns on the recording function when the rule is triggered. Mute: the phone will mute automatically when the rule is triggered. Answer: when the rule is triggered, the phone automatically answers the incoming call.
Number	Auxiliary phone number.
Type	Early: trigger execution before call establishment. Connected: trigger execution after call establishment.
Direction	For call mode, incoming/outgoing call.
Line	Set up outgoing lines.
MCAST Codec	Multicast coding.
Username	Bind the user name of the IP camera.
Password	Bind IP camera password.
URL	Video streaming information or MCAST IP address.
User Agent	Set user agent information.

12.9 Line >> Basic Settings

Set up the register global configuration.

Table 19 - Set the line global configuration on the web page

Parameters	Description
STUN Settings	
Server Address	Set the STUN server address.
Server Port	Set the STUN server port, default is 3478.
Binding Period	STUN valid time.
SIP Waiting Time	Set the timeout of STUN binding before sending SIP messages.
The TLS authentication	
TLS Certification File	Upload or delete the TLS certification file used for encrypted SIP transmission.

12.10 Line >> RTCP-XR

RTCP-XR mode is based on RFC3611 (RTP Control Extended Report), which can measure and evaluate network packet loss, delay and voice quality by sending RTCP-XR packets.

Table 20 - VQ RTCP-XR Settings

Parameters	Description
VQ RTCP-XR Settings	
VQ RTCP-XR Session Report	VQ report on whether session mode is enabled or not.
VQ RTCP-XR Interval Report	Whether to turn on Interval mode for VQ report sending.
Period for Interval Report(5~99)	The time interval at which VQ reports are sent periodically.
Warning threshold for Moslq(15~40)	When the phone calculated the Moslq value x10 below the set threshold, a warning was issued.
Critical threshold for Moslq(15~40)	When the phone calculates the Moslq value x10 below the set threshold, the critical report is issued.
Warning Threshold for Delay(10~2000)	When the one-way delay of the phone is greater than the set threshold, warning is issued.
Critical Threshold for Delay(10~2000)	When the phone computes that the one-way delay is greater than the set threshold, the critical report is issued.
Display Report Options on Phone	Whether to display the VQ report data of the last call on the phone.
Display Report Options on web	Whether to display the VQ report data for the last call through the web page.

12.11 Phone settings >> Features

Configuration phone features.

Table 21 - General function Settings

Parameters	Description
Basic Settings	
Enable Call Waiting	Enable this setting to allow user to take second incoming call during an established call. Default enabled.
Enable Call Transfer	Enable Call Transfer.
Semi-Attended Transfer	Enable Semi-Attended Transfer by selecting it
Enable 3-Way Conference	Enable 3-way conference by selecting it
Enable Auto Onhook	The phone will hang up and return to the idle automatically at hands-free mode
Auto HangUp Delay	enable Auto hangUp delay.
Auto HangUp Tone	auto HangUp Tone.
Busy HangUp Delay	When the receiving line is busy, the delay time setting for automatically returning to standby.
Ring from Headset	Enable Ring for Handset by selecting it, the phone plays ring tone from handset.
Enable Auto Headset	Enable this feature, headset plugged in the phone, user press 'answer' key or line key to answer a call with the headset automatically.
Enable Silent Mode	When enabled, the phone is muted, there is no ringing when calls, you can use the volume keys and mute key to unmute.
Disable Mute for Ring	When it is enabled, you can't mute the phone
Enable Default Line	If enabled, user can assign default SIP line for dialing out rather than SIP1.
Enable Auto Switch Line	Enable phone to select an available SIP line as default automatically
Default Ext Line	Select the default line to use for outgoing calls
Ban Outgoing	If you select Ban Outgoing to enable it, and you cannot dial out any number.
Hide DTMF	Configure the hide DTMF mode.
Enable CallLog	Select whether to save the call log.
Enable Restricted Incoming List	Whether to enable restricted call list.
Enable Allowed Incoming List	Whether to enable the allowed call list.
Enable Restricted Outgoing	Whether to enable the restricted allocation list.

List	
Enable Country Code	Whether the country code is enabled.
Country Code	Fill in the country code.
Area Code	Fill in the area code.
Enable Number Privacy	Whether to enable number privacy.
Match Direction	Matching direction, there are two kinds of rules from right to left and from left to right.
Start Position	Open number privacy after the start of the hidden location.
Hide Digits	Turn on number privacy to hide the number of digits.
Enable DTMF / Transfer	Enable DTMF / Transfer
Enable DTMF / Hold	Enable DTMF / Hold
Enable DTMF / Conference	Enable DTMF / Conference
Allow IP Call	If enabled, user can dial out with IP address.
P2P IP Prefix	Prefix a point-to-point IP call.
Caller Name Priority	Change caller ID display priority.
Search path	Select the search path.
LDAP Search	Select from with one LDAP for search.
Caller Display Type	Caller Display.
Emergency Call Number	Configure the Emergency Call Number. Despite the keyboard is locked, you can dial the emergency call number.
Restrict Active URI Source IP	Set the device to accept Active URI command from specific IP address.
Push XML Server	Configure the Push XML Server, when phone receives request, it will determine whether to display corresponding content on the phone which sent by the specified server or not.
Enable Pre-Dial	Disable this feature, user enter number will open audio channel automatically. Enable the feature, user enter the number without opening audio channel.
Enable Multi Line	If enabled, up to 10 simultaneous calls can exist on the phone, and if disabled, up to 2 simultaneous calls can exist on the phone.
Line Display Format	Custom line format: SIPn/SIPn: xxx/xxx@SIPn
Contact As Allowed List Type	Contact As AllowedList Type.
Block XML When Call	Disable XML push on call.
SIP notify	When enabled, the phone displays the information when it receives the relevant notify content.
Call Number Filter	Set call Number Filter.
Auto Resume Current	Resume automatically when the current call changes.

Call Timeout	The remote phone does not answer within the time, the local automatically hangs up.
Ring Timeout	It configures ringing time of incoming call.
Enable Push XML Auth	Enabled authentication when push xml.
Display BLF PickUp Popup	When there is an incoming call on the subscribed BLF number, Whether display the call prompt interface.
Play BLF PickUp Tone	When there is an incoming call on the subscribed BLF number, whether to play prompt tone.
Ring Type For BLF PickUp	When there is an incoming call on the subscribed BLF number, the phone plays the ring tone type.
Ring Priority	It configures the priority of incoming call.
Enable Display To Info	Display information about the to field.
Tone Settings	
Enable Holding Tone	When turned on, a tone plays when the call is held.
Enable Call Waiting Tone	When turned on, a tone plays when call waiting.
Play Dialing DTMF Tone	Play DTMF tone on the device when user pressed a phone digits at dialing, default enabled.
Play Talking DTMF Tone	Play DTMF tone on the device when user pressed a phone digits during taking, default enabled.
Auto Answer Tone	Whether enable auto answer tone.
Low Battery Tone	Whether enable battery low tone.
Ring Back Tone	Configure ring back tone.
Busy Tone	Configure busy tone.
DND Settings	
DND Option	Select to take effect on the line or on the phone or close.
Enable DND Timer	Enable DND Timer, If enabled, the DND is automatically turned on from the start time to the off time.
DND Start Time	Set DND Start Time.
DND End Time	Set DND End Time.
Intercom Settings	
Enable Intercom	When intercom is enabled, the device will accept the incoming call request with a SIP header of Alert-Info instruction to automatically answer the call after specific delay.
Enable Intercom Mute	Enable mute mode during the intercom call.
Enable Intercom Tone	If the incoming call is intercom call, the phone plays the intercom tone.
Enable Intercom Barge	Enable Intercom Barge by selecting it, the phone auto answers the intercom call during a call. If the current call is intercom call, the phone will

	reject the second intercom call
Redial Settings	
Enable Call Completion	Enable call completed to redial.
Enable Auto Redial	Enable auto redial.
Auto Redial Interval	Auto redial Interval.
Auto Redial Times	Auto redial times.
Response Code Settings	
DND Response Code	Set the SIP response code on call rejection on DND.
Busy Response Code	Set the SIP response code on line busy.
Reject Response Code	Set the SIP response code on call rejection.
Bluetooth Settings	
Enable Bluetooth	Whether to enable Bluetooth.
Bluetooth Name	Bluetooth name.
Password Dial Settings	
Enable Password Dial	Enable Password Dial by selecting it, When number entered is beginning with the password prefix, the following N numbers after the password prefix will be hidden as *, N stand for the value which you enter in the Password Length field. For example: you set the password prefix is 3, enter the Password Length is 2, then you enter the number 34567, it will display 3**67 on the phone.
Encryption Number Length	Configure the Encryption Number length.
Password Dial Prefix	Configure the prefix of the password call number.
Bluetooth Settings	
Enable Bluetooth	Enable / disable the bluetooth.
Bluetooth Name	The name of the device when the bluetooth is enabled.
Notification Popups	
Display Missed Call Popup	No incoming call popup prompt after opening, no popup prompt when closing, open by default.
Display Voice Mail Popup	Display voice mail popup.
Display SMS Popup	There is popup prompt for unread messages after opening, and there is no popup prompt when closing. It is opened by default.
Display Other Popup	When the handle is not hung back after opening, registration fails, IP acquisition fails, Tr069 connection fails and other abnormalities, there will be popup prompt when it is opened; otherwise, there will be no prompt when it is closed, and it will be opened by default.

12.12 Phone settings >> Media Settings

Change voice Settings.

Table 22 - Voice settings

Parameters	Description
Codecs Settings	Select enable or disable voice encoding: G.711A/U, G.722, G.729AB, G.726-16, G726-24, G726-32, G.726-40, G723.1, iLBC, Opus
Video Codecs	Select enable or disable video encoding:H264
Noise Reduction	
A.I Noise Reduction Technology	Whether to enable A.I noise reduction technology.
Acoustic Shield	Whether to enable acoustic shield.
Media Settings	
Handset Volume	Set the handset volume, the value must be 1~9.
Default Ring Type	Configure default ringtones. If no special ringtone is set for the phone number, the default ringtone will be used.
Speakerphone Volume	Set the hands-free volume to 1-9.
Headset Ring Volume	Set the volume of the headset ringtone to 1~9.
Headset Volume	Set the volume of the headset to 1~9.
Speakerphone Ring Volume	Set the volume of hands-free ringtone to 1~9.
DTMF Payload Type	Enter the DTMF payload type, the value must be 96~127.
Headset Mic Gain	Set the Headset's radio volume gain to fit different models of headsets.
Handset Mic Gain	Set the Handset's radio volume gain to fit different models of headsets.
Handfree Mic Gain	Set the Handfree's radio volume gain to fit different models of headsets.
Opus Payload Type	Set Opus load type, range 96~127.
Opus Sample Rate	Set Opus sampling rate, including opus-nb (8KHz) and opus-wb (16KHz).
iLBC Payload Type	Set the iLBC Payload Type, the value must be 96~127.
iLBC Payload Length	Set the iLBC Payload Length.
Enable VAD	Whether voice activity detection is enabled.
Enable Voice Mail Tone	It enables or disables the IP phone to play voice message tone when dialing.
Enabled Vibrate	Whether to enable vibrate.
Video Bit Rate	It defines the maximum bit rate for video transmission.
Video Resolution	It sets the resolution (frame size) for the video stream.
H.264 Payload Type	It specifies the RTP payload type number for H.264 video packets.
Video Display Mode	It configures how the video stream is displayed, such as aspect ratio

	handling.
RTP Control Protocol(RTCP) Settings	
CNAME user	Set CNAME user
CNAME host	Set CNAME host
RTP Settings	
RTP keep alive	Hold the call and send the packet after 30 s.
RTP Relay	Set RTP Relay.
Alert Info Ring Settings	
Value	Set the value to specify the ring type.
Line	Set the line to specify the ring type.
Ring Type	Set ring type.

12.13 Phone settings >> MCAST

This function allows simple and convenient announcement delivery to every multicast member. By configuring a multicast key on the phone, multicast RTP streams can be sent to a preconfigured multicast address without involving any SIP signaling.

Users can configure multicast listening addresses on the phone to monitor and play RTP streams sent to these addresses, also without SIP signaling. A maximum of 10 multicast listening addresses can be configured.

Table 23 - Multicast parameters

Parameters	Description
SIP Priority	Define the priority of the active call, 1 is the highest priority, 10 is the lowest.
Intercom Priority	Set intercom priority.
Enable Page Priority	The voice call in progress shall take precedence over all incoming paging calls.
PTT MCAST Standby Sync	Set PTT MCAST standby sync.
Enable Prio Chan	Once enabled, only the same port and channel can be connected. Channel 24 is the priority channel, which takes precedence over channels 1-23; a channel value of 0 indicates that no channel is used.
Mcast Listening Renew Time	Set the waiting duration for re-listening to Mcast.
Enable Emer Chan	Once enabled, channel 25 will have the highest priority.
Name	Listened multicast server name
Host: port	Listened multicast server's multicast IP address and port.
Channel	Set channel.0-25

12.14 Phone settings >> Action

Action URL

Action URL is used for IPPBX systems to submit phone events.

12.15 Phone settings >> Time/Date

The user can configure the time Settings of the phone on this page.

Table 24 - Time&Date settings

Parameters	Description
Network Time Server Settings	
Time Synchronized via SNTP	Enable time-sync through SNTP protocol.
Time Synchronized via DHCP	Enable time-sync through DHCP protocol.
Time Synchronized via DHCPv6	It enables or disables the IP phone to update time with the offset time offered by the DHCP server.
Primary Time Server	Set primary time server address.
Secondary Time Server	Set secondary time server address, when primary server is not reachable, the device will try to connect to secondary time server to get time synchronization.
Time Zone	Select the time zone.
Resync Period	Time of re-synchronization with time server.
Dynamically Display Time	Whether to enable dynamically display time.
Time/Date Format	
12-Hour clock	Set the time display in 12-hour mode.
Time/Date Format	Select the time/date display format.
Daylight Saving Time Settings	
Location	Choose your local, phone will set daylight saving time automatically based on the local.
DST Set Type	Choose DST Set Type, if Manual, you need to set the start time and end time.
Fixed Type	Daylight saving time rules are based on specific dates or relative rule dates for conversion. Display in read-only mode in automatic mode.
Offset	The offset minutes when DST started.
Month Start	The DST start month.
Week Start	The DST start week.
Weekday Start	The DST start weekday.
Hour Start	The DST start hour.

Minute Start	The DST start minute.
Month End	The DST end month.
Week End	The DST end week.
Weekday End	The DST end weekday.
Hour End	The DST end hour.
Minute End	The DST end minute.
Manual Time Settings	You can set your time manually.

12.16 Phone settings >> Time Plan

Time Plan (time management) settings can set a time point or a time period. The time point is to perform an action at a certain time, and the time period is to perform an action for a certain period of time.

Time Plan:

Name:

Type:

Repetition period:

Start Date:

End Date:

Effective time:

Time Plan List:

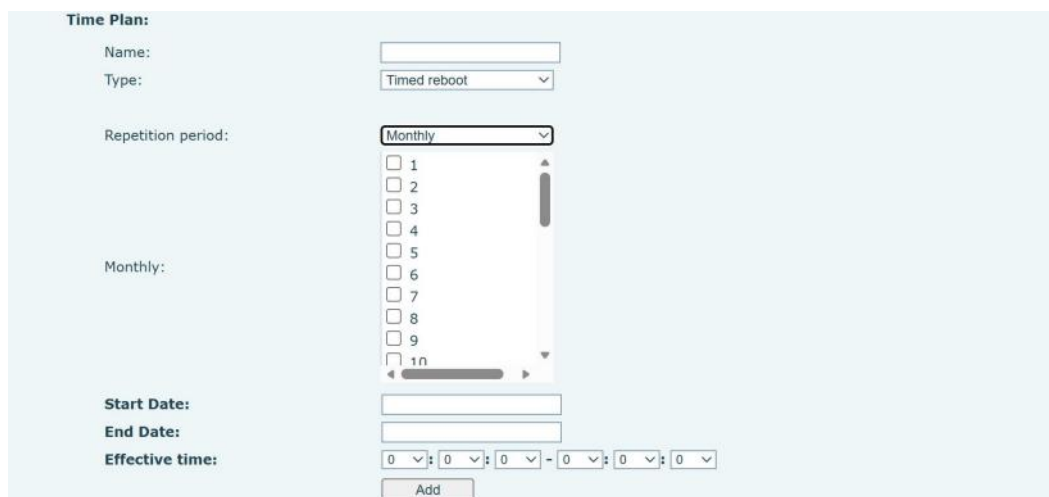
☐	Index	Name	Type	Special configure	Repetition period	Start Date	End Date	Effective time
Delete								

Picture 145 - Time Plan (1)

Table 25 - Time Plan

configure	Value	Description
Time plan Type	1: Timed reboot 2: Timed upgrade 3: Timed forward 4: Timed config	Type , Action performed at a time point/time period.
Repetition period	0: No repetition 1: Daily 2: Weekly 3: Monthly	repeat type
in weeks	0-6: Sunday-Saturday, supports multiple separated by ";" 1-31: 1-31 day	When the repetition type is daily/non-repeating, the value is empty
in days	xx:xx-xx:xx	start time - end time period

When the Time Plan type is selected as timed forwarding, the web page will prompt to enter the forwarding number and forwarding line, as shown in the figure.



Picture 146 - Time Plan (2)

Forwarding Number: Configure the forwarding number to forward to the number within the set time period.

Line: Forward the specified line, when the line is set to a certain line, it will only take effect for this line.

1. Timed forwarding rules:

1) When there is forwarding under the line, the forwarding number under the line is used; when there is no forwarding number under the SIP line, when there is an incoming call within the time period set by the scheduled forwarding, the phone will be forwarded to the specified scheduled forwarding number; when outside the time period, no forwarding is performed. That is, the priority Line>Time Plan.

2) All scheduled forwarding types are unconditional forwarding.

12.16.1 Repeat Period Select Daily

Select daily as the repetition period, and enter any time in the date format from 00:00 to 23:59 in the effective time input box.

The first and third input boxes only allow input of any integer from 00 to 23, and 0 is automatically added before inputting an integer less than 10.

The second and fourth input boxes only allow input of any integer from 00 to 59, and 0 is automatically added before inputting an integer less than 10.

Repetition period: Daily

Start Date:

End Date:

Effective time: 0:0:0 - 0:0:0

Picture 147 - Time Plan (3)

12.16.2 Repeat Period Select Weekly

Day of the week selection box, check it to take effect.

The final effective time is the combination of the day of the week and the set time.

Repetition period: Weekly

Weekly: ☐ Sunday
☐ Monday
☐ Tuesday
☐ Wednesday
☐ Thursday
☐ Friday
☐ Saturday

Start Date:

End Date:

Effective time: 0:0:0 - 0:0:0

Picture 148 - Time Plan (4)

12.16.3 Time Plan List

All configurations submitted after the configuration is submitted are displayed in a list, and the order is sorted by week (day, Monday, Tuesday...), and if the week is the same, it is sorted by time (time from small to large). The function sequence is restarted first and then upgraded.

Time Plan List:

☐	Index	Name	Type	Special configure	Repetition period	Start Date	End Date	Effective time
☐	1		Timed reboot		Daily	2025-08-13	2025-08-31	1:2:0-3:3:0
☐	2		Timed reboot		No repetition	2025-08-18		9:0:0-18:0:0

Picture 149 - Time Plan (5)

12.16.4 Delete

Check the box next to "Index" to select or deselect all configuration items in the list. Check the box before each number to select or deselect the corresponding configuration row. Click **[Delete]** to remove the checked configurations from the list; deleted configurations will become invalid immediately.

Time Plan List: ?

☐	Index	Name	Type	Special configure	Repetition period	Start Date	End Date	Effective time
☒	1		Timed reboot		Daily	2025-08-13	2025-08-31	1:2:0-3:3:0
☒	2		Timed reboot		No repetition	2025-08-18		9:0:0-18:0:0

Picture 150 - Time Plan (6)

12.17 Phone settings >> Tone

This page allows users to configure a phone prompt.

The user can either select the country area or customize the area. If the area is selected, it will bring out the following information directly. If the user choose to customize the area, then the user can modify the key tone, call back tone and other information.

Tone Settings

Select Your Tone: United States ?

Dial Tone: 350+440/0 ?

Ring Back Tone: 440+480/2000,0/4000 ?

Busy Tone: 480+620/500,0/500 ?

Congestion Tone: ?

Call waiting Tone: 440/300,0/10000,440/300,0/10000,0/0 ?

Holding Tone: ?

Error Tone: ?

Stutter Tone: ?

Information Tone: ?

Dial Recall Tone: 350+440/100,0/100,350+440/100,0/100,350+440/100,0/100,350+440/0 ?

Message Tone: ?

Howler Tone: ?

Number Unobtainable Tone: 400/500,0/6000 ?

Warning Tone: 1400/500,0/0 ?

Record Tone: 440/500,0/5000 ?

Auto Answer Tone: ?

Picture 151 - Tone settings on the web

12.18 Phone settings >> Alarm

Please refer to [9.1 Alarm](#).

12.19 Phone settings >> Advanced

User can configure the advanced configuration settings in this page.

- Screen Configuration.
 - Enable Energy Saving
 - Backlight Time
 - Timeout to Sleep
- UI Preference

Set font, color, and screensaver display

- LCD Menu Password Settings.

The password is 123 by default.

- Keyboard Lock Settings.
- Greeting Words

The greeting message will display on the LCD when the device is idle, which is limited to 16 characters. The default chars are "VOIP PHONE".

12.20 Phonebook >> Contacts

User can add, delete, or edit contacts in the phonebook in this page. User can browse the phonebook and sorting it by name, phones, or filter them out by group.

To add a new contact: After clicking "Add New Contact", enter the contact information and click the "Confirm" button to complete the addition.

To Edit a contact: Click the **[Edit]** button at the end of the row corresponding to the contact, modify the information in the edit field, then click the **[Confirm]** button to finish editing.

To delete one or more contacts, select the checkboxes next to the serial numbers of the desired contacts and click the **[Delete]** button. Click **[Delete All]** to remove all contacts in the phone book.

Users can also add multiple contacts to a group. First, select the checkboxes next to the corresponding rows of the contacts to be added to the group. Find the **[Add to Group]** button at the bottom of the contact list, select the target group name from the drop-down list in front of the button, and then click the **[Add to Group]** button to add the selected contacts to the chosen group.

Similarly, users can select multiple contacts and add them to the allowed call list/blocked call list by clicking the **[Add to Allowed List]** / **[Add to Blocked List]** button.

12.21 Phonebook >> cloud contacts

cloud contacts

User can configure up to 4 cloud contacts. Each cloud contacts must be configured with an URL where an XML phonebook is stored. The URL may be based on HTTP/HTTPS or FTP protocol with or without authentication. If authentication is required, user must configure the username and password.

To configure a cloud contacts, the following information should be entered,

Cloud Contacts name (must)

Cloud Contacts URL (must)

Authentication username (optional)

Authentication password (optional)

LDAP Settings

The cloud contacts allows user to retrieve contact list from a LDAP Server through LDAP protocols.

User must configure the LDAP Server information and Search Base to be able to use it on the device. If the LDAP server requests an authentication, user should also provide username and password.

To configure a LDAP phonebook, the following information should be entered,

- Display Title (must)
- LDAP Server Address (must)
- LDAP Server Port (must)
- Search Base (must)
- Access username (optional)
- Access password (optional)

Web page preview

The phone web page supports the preview of Internet phone directory and contacts

- After setting up the XML directory or LDAP,
- Select **[Phonebook] >> [cloud contacts] >> [cloud contacts]** to select the type.
- Click the set XML/LDAP to download the contact for browsing.

Picture 152 - Web cloud phone book Settings

12.22 Phonebook >> Call List

■ Restricted Incoming Calls:

It is similar to a Blocked List. Add the number to the Blocked List, and the user will no longer receive calls from the stored number until the user removes it from the list.

Users can add specific Numbers to the Blocked List or add specific prefixes to the Blocked List to block calls with all Numbers with this prefix.

■ Allowed Incoming Calls:

When DND is enabled, the incoming call number can still be called.

■ Restricted Outgoing Calls:

Numbers added to the outgoing call restriction list cannot be dialed initiatively by the phone until they are removed from the list.

12.23 Phonebook >> Web Dial

Users can perform call origination, answer and hang-up operations via the web page here.

12.24 Phonebook >> Advanced

Users can export the local phone book in XML, CSV, and VCF format and save it on the local computer.

Users can also import contacts into the phone book in XML, CSV, and VCF formats.

Attention! If the user imports the same phonebook repeatedly, contacts with identical names and numbers will not be added again. If contacts have the same name but different numbers, new entries will be created and added.

Users can delete groups or add new groups on this page. Deleting a contact group does not delete contacts in that group.

12.25 Call Log

The user can browse the complete call record in this page. The call record can be sorted by time. Call number, contact name or line, and the call record can be screened by call record type (incoming call, outgoing call, missed call, forward call).

The user can also save the number in the call record to his/her phone book or add it to the Blocked List/whitelist.

Users can also dial the web page by clicking on the number in the call log, the phone will call the number clicked automatically.

12.26 Function Key >> Function Key

One-key transfer Settings: establish new call, blind transfer, attention-transfer, one-key three-way, Play DTMF.

The device provides multiple user-defined shortcut keys, and the user can configure each shortcut key on the web page.

Table 26 - Function Key configuration

Parameters	Description
Memory Key	BLF (New Call/BXFE /AXFER): It is used to prompt user the state of the subscribe extension, and it can also pick up the subscribed number, which help user monitor the state of subscribe extension (idle, ringing, a call). There are 3 types for one-touch BLF transfer method. p.s. User should enter the pick-up number for specific BLF key to fulfill the pick-up operation. Presence: Compared to BLF, the Presence is also able to view whether the user is online. Note: You cannot subscribe the same number for BLF and Presence at the same time

	Speed Dial: You can call the number directly which you set. This feature is convenient for you to dial the number which you frequently dialed. Intercom: This feature allows the operator or the secretary to connect the phone quickly; it is widely used in office environments. Voice Mail: Used to configure voice mail numbers and lines. When the indicator light is on, it indicates that there is new voice information. Call Park: Used to switch to other devices or locations during a call. Call forward: After the configuration, you can press this key to enable Call Forward Unconditional to a specified number when the device is idle. When the device receives an incoming call, press this key to forward the call to a specified number.
BLF List Key	This function depends on the Broadsoft server. Add the numbers that need to be subscribed to in the server to the group, use the URL address of the group to subscribe to all the numbers in the group, and display the corresponding number, name, and status on the indicator light of the BLF List key. The BLF List is automatically issued by the server based on configuration, without the need for users to manually configure values and names.
Key Event	User can select a key event as a shortcut to trigger. For example: Voice Mail / Do - not - disturb / Call hold / Call transfer / Phonebook / Redial / Pickup / Join / Call forward / Call Logs / Flash / Memo / Headset / Release / Lock phone / SMS / Call Back / Hide DTMF / Intercom / Group Listening / Prefix / Hot Desking / Agent / End / Disposition / Escalate / Trace / Handfree / Answer Key / Private Hold / Local Contacts / LDAP Group / XML Group / Broadsoft Group / Record / Auto Headset / Acoustic Shield / Noise Reduction / DSS Key Page1 / DSS Key Page2.
DTMF	Press this key during a call to send the configured values in sequence to the remote end.
URL	Access the configured remote URL address, which can be set to an XML phone book address, etc.
Multicast Paging	Configure the multicast address and audio codec. User presses the key to initiate the multicast.
XML browser	Users can set the DSS Key for specific URL download and other operations.
PTT	Users can set PTT one key call to realize the function of pressing and holding the call and releasing the hang up.

12.27 Function Key >> PTT Key

Users can configure the red PTT key on the side of the phone on this page. The available options are the same as above.

12.28 Function Key >> Softkey

The User Settings mode and display style, display page.

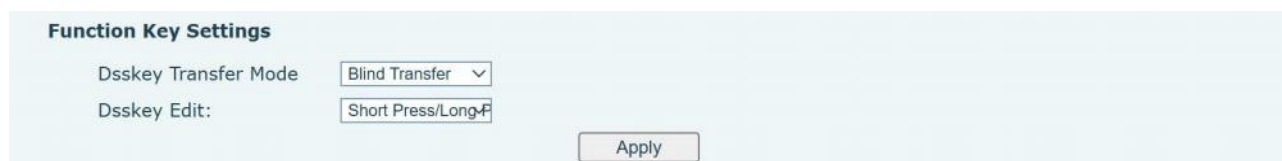
Table 27 - Softkey configuration

Parameter	Description
Softkey Mode	
Softkey mode	Disabled and More, Default is More
Softkey Style	
Softkey display style	Softkey Exit on Left or Right
Screen	
Call Dialer	Redial / 2aB / Delete / Exit / Call Back / Dial / Join / MWI / Local Contacts / Pickup / Call Log / Missed / Clear / In / Dialed / Pause / Next line / Prev line / Headset / Audio / Video / Remote XML / DSS Key
Conference(Conf)	Hold / Split / End /Release / Mute / DSS Key / Headset
Desktop	Call Log / Menu / Local Contacts / DND / Prev Account / Next Account / Blocked List / Call Back / Call Forward / Locked / Memo / Missed / MWI / Dialed / Reboot / Redial / Remote XML / SMS / Headset / Status / DSS Key / In
Divert Dialer	Redial / 2aB / Delete / Exit / Forward / Local Contacts / Call Log / Clear / Missed / Dialed / Headset / Video / Audio / Remote XML / DSS Key
Ending	Redial / End / Headset / Release / DSS Key
Predictive Dialer	Dial / 2aB / Delete / Exit / Call Back / Local Contacts / Redial / Pickup / MWI / Join / Call Log / Release / Missed / Pause / Dialed / Headset / Video / Audio / Remote XML / DSS Key / In / Next line / Prev line
Ringing	Answer / Forward / Reject / Mute / Release / Headset / Video / Audio / DSS key
Talking Audio	Hold / Transfer / Conference / End / Mute / Release / New Call / Local Contacts / Listen / Call Log / Next call / Prev call / Private / Headset / Video / Audio / DSS Key
Talking Video	Hold / Switch / Transfer / End / Conference(Conf) / Mute / NewCall(New) / Release / Local Contacts / Listen / RTP / AcousticShield / Noise Reduction / Open / CallLog / Next Call(Next) / Prev.Call(Prev.) / PrivateHold(PriHold) / Headset / Video / Audio / DsskeyPage1 / DsskeyPage2 / Soft Dsskey
Transfer Alerting	End / Transfer / Headset / Release / DSS Key
Transfer Dialer	Redial / Delete / Exit / 2aB / Dial / Local Contacts / Transfer / Call Log / Clear / Missed / Dialed / Pause / Headset / Video / Audio / Remote XML / DSS Key
Trying	End / Release / Headset / DSS Key
Waiting	Hold / Transfer / Conference / End / Answer / Forward / Mute / Next call / New call / Prev call / Reject / Release / Headset / Listen / Video / Audio / DSS Key

12.29 Function Key >> Advanced

■ Function Key Settings

One key transfer: for example, set the memory key 4370. Press the memory key when talking with 4374 to decide whether to call 4370 or transfer 4374 to 4370.

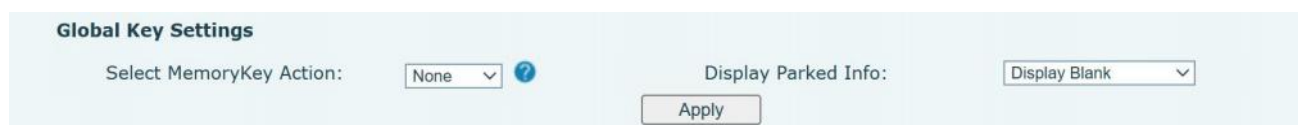


The screenshot shows the 'Function Key Settings' window. It contains two settings: 'Dsskey Transfer Mode' set to 'Blind Transfer' and 'Dsskey Edit' set to 'Short Press/Long-P'. An 'Apply' button is located at the bottom right.

Picture 153 - Function Key Settings

■ Global Key Settings

Select memory key function: for example, the phone set the memory key value to 4370. When 4370 calls, press this key to hold the call or hang up.



The screenshot shows the 'Global Key Settings' window. It contains two settings: 'Select MemoryKey Action' set to 'None' and 'Display Parked Info' set to 'Display Blank'. An 'Apply' button is located at the bottom center.

Picture 154 - Global Key Settings

■ Programmable key Settings

Please refer to [Table 27 - Softkey configuration](#).

12.30 Application >> Doorphone Settings

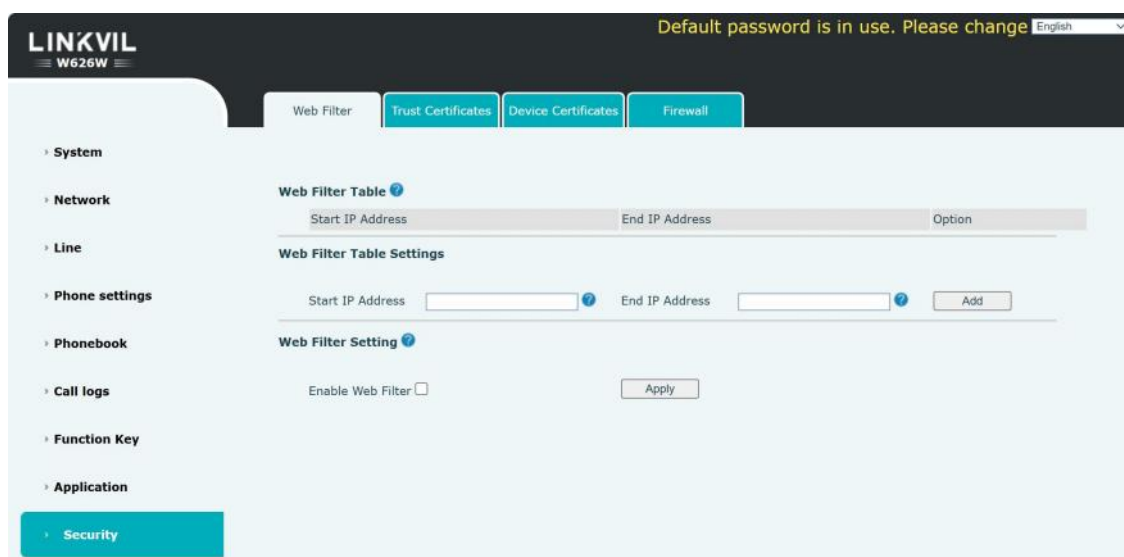
Please refer to [9.4 One-Click to Open Door](#).

12.31 Application >> Manage Recording

See [9.5Record](#) for details of recording.

12.32 Security >> Web Filter

The user can set up a configuration management phone that allows only machines with a certain network segment IP access.



Picture 155 - Web Filter settings



Picture 156 - Web Filter Table

Adding and removing IP segments are accessible. Configure the starting IP address within the start IP, end the IP address within the end IP, and click **[Add]** to submit to take effect. A large network segment can be set, or it can be divided into several network segments to add. If the user wants to delete, select the initial IP of the network segment to be deleted from the drop-down menu, and then click **[Delete]** to take effect.

Enable web page filtering: configure enable/disable web page access filtering; Click the **[apply]** key to take effect.

Note: if the device you are accessing is in the same network segment as the phone, please do not configure the filter segment of the web page to be outside your own network segment, otherwise you will not be able to log in the web page.

12.33 Security >> Trust Certificates

Set whether to open license certificate and general name validation, select certificate module. You can upload and delete uploaded certificates.

Permission Certificate

Permission Certificate ?

Common Name Validation ?

Certificate mode ?

Import Certificates ?

Load Server File

Certificates List ?

Index	File Name	Issued To	Issued By	Expiration	File Size
					Delete

Picture 157 - Certificate of settings

12.34 Security >> Device Certificates

Select the device certificate as the default and custom certificate.

You can upload and delete uploaded certificates.

Device Certificates ?

Device Certificates (existence)

Import Certificates ?

load Device file

Certification File ?

File Name	Issued To	Issued By	Expiration	File Size
				Delete

Picture 158 - Device certificate setting

12.35 Security >> Firewall

Picture 159 - Network firewall Settings

The user can set whether to enable the input through this page, output firewall and set the firewall input and output rules. Using these Settings can prevent some malicious network access, or restrict internal users access to some resources of the external network, which can improve security.

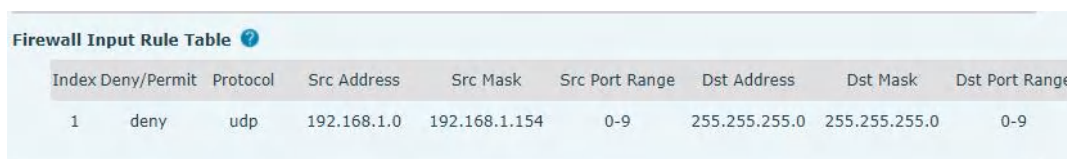
Firewall rule set is a simple firewall module. This feature supports two types of rules: input rules and output rules. Each rule is assigned an ordinal number, allowing up to 10 for each rule. Considering the complexity of firewall Settings, the following is an example to illustrate:

Table 28 - Network Firewall

Parameter	Description
Enable Input Rules	Indicates that the input rule application is enabled.
Enable Output Rules	Indicates that the output rule application is enabled.
Input/Output	To select whether the currently added rule is an input or output rule.
Deny/Permit	To select whether the current rule configuration is disabled or allowed;
Protocol	There are four types of filtering protocols: TCP UDP ICMP IP.
Src Port Range	Filter port range
Src Address	Source address can be host address, network address, or all addresses 0.0.0.0; It can also be a network address similar to *.*.*.0, such as: 192.168.1.0.
Dst Address	The destination address can be either the specific IP address or the full address 0.0.0.0; It can also be a network address similar to *.*.*.0, such as: 192.168.1.0.
Src Mask	Is the source address mask. When configured as 255.255.255.255, it means that the host is specific. When set as 255.255.255.0, it means that a network segment is filtered.
Dst Mask	Is the destination address mask. When configured as 255.255.255.255, it means

	the specific host. When set as 255.255.255.0, it means that a network segment is filtered.
--	--

After setting, click **[Add]** and a new item will be added in the firewall input rule, as shown in the figure below:

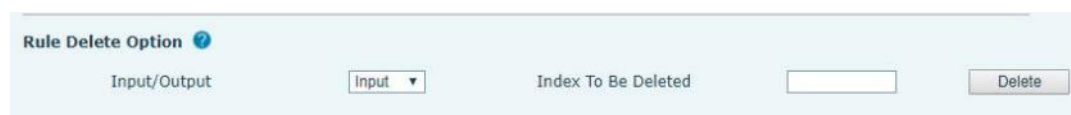


Index	Deny/Permit	Protocol	Src Address	Src Mask	Src Port Range	Dst Address	Dst Mask	Dst Port Range
1	deny	udp	192.168.1.0	192.168.1.154	0-9	255.255.255.0	255.255.255.0	0-9

Picture 160 - Firewall Input rule table

Then select and click the key **[Apply]**.

In this way, when the device is running: ping 192.168.1.118, the packet cannot be sent to 192.168.1.118 because the output rule is forbidden. However, the other IP of the ping 192.168.1.0 network segment can still receive the response packet from the destination host normally.



Rule Delete Option

Input/Output Input Index To Be Deleted Delete

Picture 161 - Delete firewall rules

Select the list you want to delete and click **[Delete]** to delete the selected list.

12.36 Device Log >> Device Log

You can grab the device log, and when you encounter an abnormal problem, you can send the log to the technician to locate the problem. See [13.7 Get log information](#).

13 Trouble Shooting

When the phone is not in normal use, the user can try the following methods to restore normal operation of the phone or collect relevant information and send a problem report to Fanvil technical support mailbox.

13.1 Get Device System Information

Users can get information by pressing **[OK]** key in phone's standby mode to enter menu >> **[Status]** option in the phone. The following information will be provided:

The network information, Equipment information (model, software and hardware version), The account information, TR069, RTP.

13.2 Power on/off

Boot up operation

- Long press the **[Red On-hook key]** until it boots up.
- Connect the charging base and it will automatically boot up.

Shutdown operation

- Shutdown: Long press the **[Red On-hook key]** to pop up a shutdown window and select whether to shut down.
- Automatic shutdown when low battery: When the battery is about to run out, it prompts "Low battery".
The device automatically shuts down when the battery is depleted.

13.3 Reboot Device

Users can reboot the device from soft-menu, press **[OK]** in phone's standby mode to enter menu >> **[Basic]** >> **[Reboot System]**, and confirm the action by **[OK]**. Or long press the **[Red On-hook key]** to reboot.

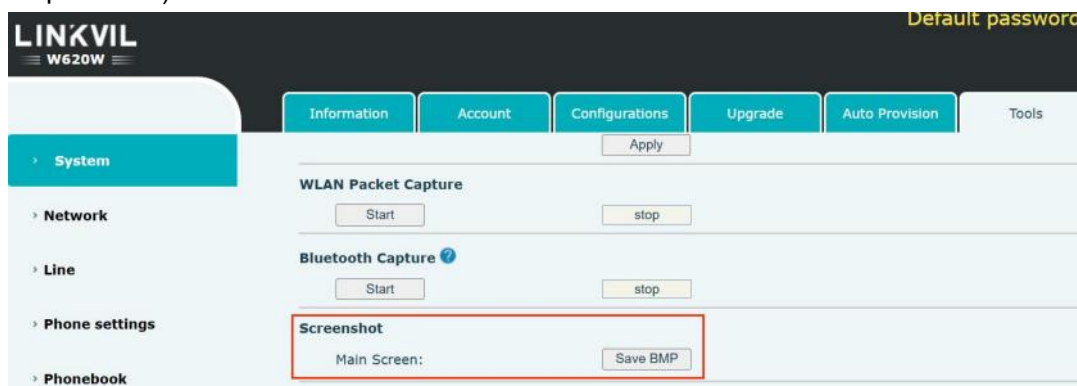
13.4 Reset Device to Factory Default

Resetting Device to Factory Default will erase all the user's configuration, preference, database and profiles on the device and restore the device back to the state as factory default.

To perform a factory default reset, user should press **[OK]** key in phone's standby mode to enter menu >> **[Advanced]**, and then input the password to enter the interface. Then choose **[Factory Reset]** and confirm the action by **[OK]** or long press the OK key on the standby interface of W626W. The device will be rebooted into a clean factory default state.

13.5 Screenshot

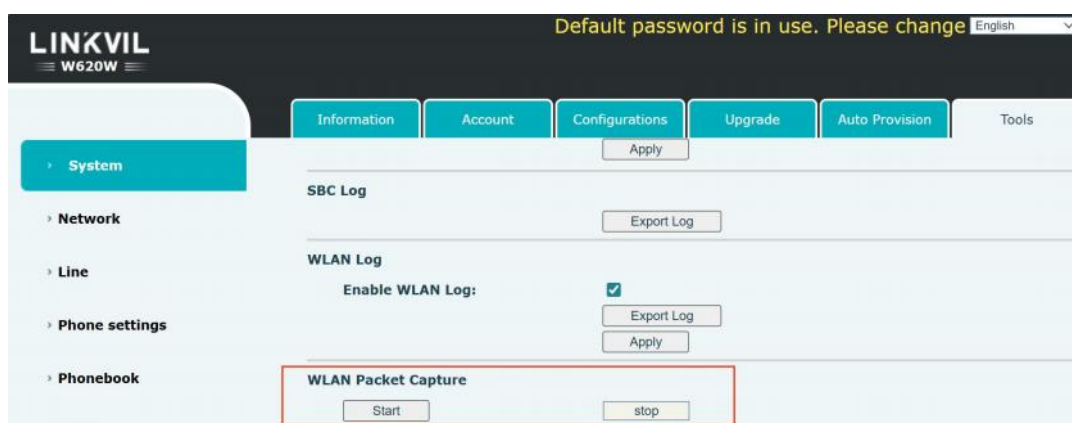
If there is a problem with the phone, the screenshot can help the technician locate the function and identify the problem. In order to obtain screen shots, log in the phone web page **[System]** >> **[Tools]**, and you can capture the pictures of the main screen and the secondary screen (you can capture them in the interface with problems).



Picture 162 - Screenshot

13.6 Network Packets Capture

Sometimes it is helpful to dump the network packets of the device for issue identification. To get the packets dump of the device, user needs to log in the device web portal, open page **[System]** >> **[Tools]** and click **[Start]** in “Network Packets Capture” section. A pop-up message will be prompt to ask user to save the capture file. User then should perform the relevant operations such as activating/deactivating line or making phone calls and click **[Stop]** key in the web page when operation finished. The network packets of the device during the period have been dumped to the saved file.



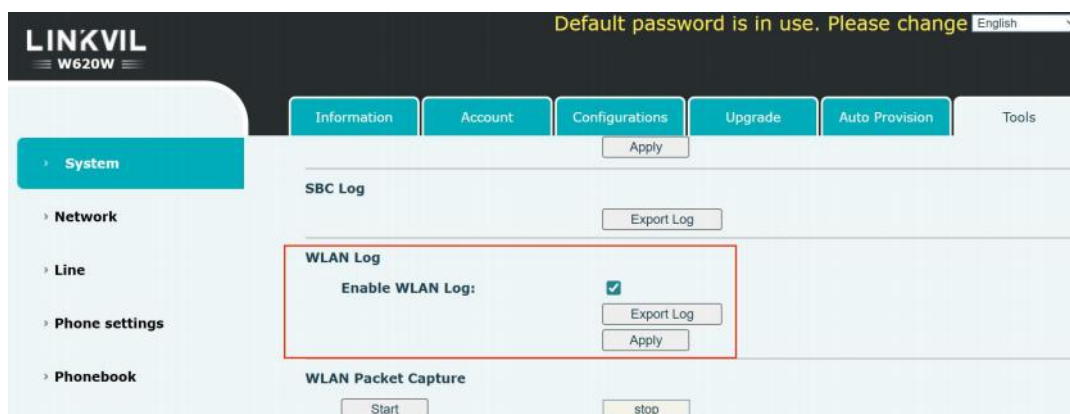
Picture 163 - Web capture

User may examine the packets with a packet analyzer or send it to Fanvil support mailbox.

13.7 Get Log Information

Log information is helpful when encountering an exception problem. In order to get the log information of

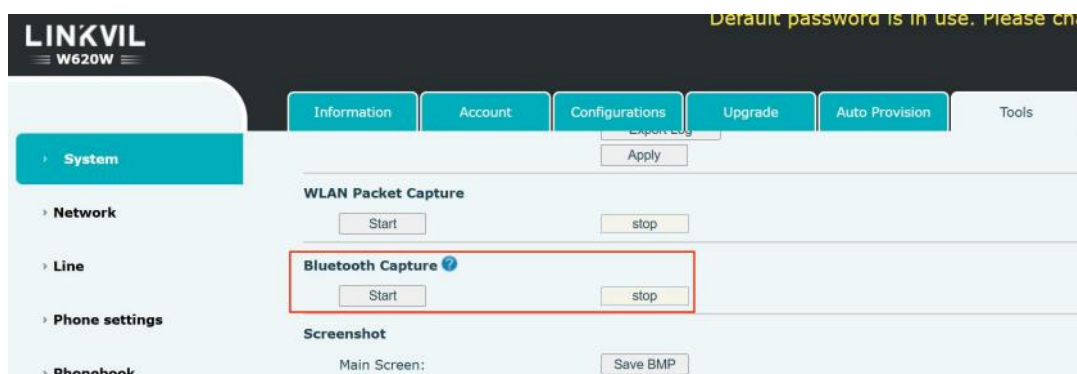
the phone, the user can log in the phone web page, open the page **[System]** >> **[Tools]** >> **[WLAN Log]**, check **[Enable WLAN Log]**, click **[Export Log]** to export log.



Picture 164 - Get WLAN Log

13.8 Get Bluetooth Data Packet

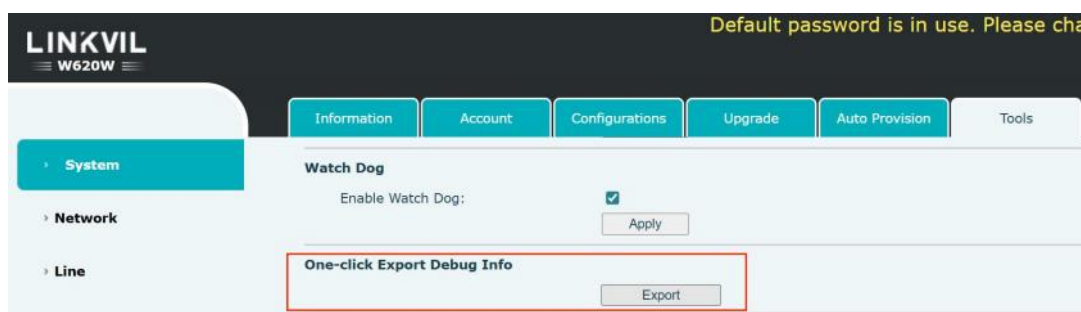
Bluetooth data packet is helpful when encountering an exception problem. In order to get the bluetooth data packet of the phone, the user can log in the phone web page, open the page **[Bluetooth Capture]**, click the **[Start]** key, follow the steps of the problem until the problem appears, and then click the **[End]** key, **[Save]** to local analysis or send the log to the technician to locate the problem.



Picture 165 - Get Bluetooth data packet

13.9 One-click Export Debug Info

Users can log in to the phone's web interface, open the "One-Click Export of Debug Info" page, and click "Export" to export debugging information with one click.



Picture 166 - One-click export debug info

13.10 Common Trouble Cases

Table 29 - Trouble Cases

Trouble Case	Solution
Device could not boot up	1. First consider the situation that the battery has been exhausted, and charge the device. Please use the charger configured at the time of purchase when charging. 2. If you see the phone enter "POST mode", it means that the phone system is damaged. Please contact technical support to help you restore the phone system.
Device could not register to a service provider	1. Please check whether the phone is connected to the network. The phone needs to be connected to a wireless network, and check whether the wireless icon is normal. If the icon shows  or , the network is unavailable, please check the network facilities. 2. Please check whether the phone has an IP address. Check the system information. If the IP address is Negotiating..., it means that the phone has not obtained an IP address. Please check whether the network configuration is correct. 3. If the network connection is good, please check your line configuration again. If all configurations are correct, please contact your service provider for support, or follow the instructions in "13.6 Network Packets Capture" to obtain a registered network packet and send it to the manufacturer's support mailbox to help analyze the problem.
No Audio or Poor Audio in Handset	1. Please check whether the headset is connected, the headset icon  is displayed in the upper right corner. After connecting the headset, the earpiece channel cannot be used; after unplugging the headset, the earpiece channel can be used. 2. At this point the network bandwidth and latency may not be suitable for audio calls.

Poor Audio or Low Volume in Headset	1. Please use the Headset provided by manufacturer. 2. The network bandwidth and delay may be not suitable for audio call at the moment.
Audio is chopping at far-end in Hands-free speaker mode	This is usually due to loud volume feedback from speaker to microphone. Please lower down the speaker volume a little bit, the phenomenon will be gone.