



# DOOR PHONE ADMIN GUIDE

Applicable Models: R28/R27/R20K

# About This Manual

Thank you for choosing Akuvox R28 series door phone. This manual is intended for the administrators who need to properly configure the door phone. This manual applies to 28.30.3.111 version, and it provides all the configurations for the functions and features of Akuvox door phone. Please visit [Akuvox web](#) or consult technical support for any new information or latest firmwares.

# Introduction of Icons and Symbols



## **Warning:**

- **Always abide by this information in order to prevent the persons from injury.**



## **Caution:**

- **Always abide by this information in order to prevent the damages to the device.**



## **Note:**

- **Informative information and advice from the efficient use of the device.**

## **Related Documentation**

You are advised to refer to the related documents for more technical information via the link below:

**<https://knowledge.akuvox.com>**

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# 1. Product Overview

The security that comes with being able to control who comes into your building along with the ability to verbally and visually confirm their identity is immeasurable. Akuvox R28A is a SIP-compliant, hands-free and video(optional) door phone. It can be connected with Akuvox indoor monitors for remote access controlling and monitoring. Users can communicate with visitors via audio and video calls, and unlock the door if they need. The door phone enables you to easily monitor an entrance door or gate and gives you peace of mind knowing that your facility is more secure.

## 2. Change Log

The change log will be updated here along with the changes in the new software version.

### 3. Model Specification

Model & Feature	R28A
	
<b>Button</b>	Physical Numeric Keypad
<b>Housing Material</b>	Aluminum
<b>Camera</b>	2 Mega pixels, automatic lighting
<b>Relay In</b>	3
<b>Relay Out</b>	3
<b>RS485</b>	√
<b>PoE</b>	√
<b>RAM</b>	128MB
<b>ROM</b>	16MB
<b>Card Reader</b>	√
<b>IP Rating</b>	IP65
<b>IK Rating</b>	X
<b>Wall Mounting</b>	√
<b>Flush Mounting</b>	√
<b>Wall Mounting Dimension</b>	280x130x38mm
<b>Flush Mounting Dimension</b>	280x130x68mm

## 4. Introduction to Configuration Menu

- **Status:** this section gives you basic information such as product information, Network Information, and account information, etc.
- **Account:** this section concerns SIP account, SIP server, proxy server, transport protocol type, audio&video codec, DTMF, session timer, etc.
- **Network:** this section mainly deals with DHCP&Static IP setting, RTP port setting, and device deployment, etc.
- **Intercom:** this section covers Intercom settings, Call Log, etc.
- **Surveillance:** this section covers Motion Detection, RTSP, MJPEG, ONVIF, Live stream.
- **Access Control:** this section covers Input control, Relay, Card settings, Face Recognition setting, Private PIN Code, Wiegand connection, etc.
- **Tenants:** this section involves Tenants management and Dial Plan.
- **Device:** this section includes Light settings, tab&button display, LCD settings and Voice settings.
- **Settings:** this section includes Time&language, Action settings, Door settings, Schedule for access control.
- **Upgrade:** this section covers Firmware upgrade, device reset&reboot, configuration file auto-provisioning, fault Diagnosis.
- **Security:** this section is for Password modification.
  
- **Mode selection :**
  1. **Discovery mode:** It is a plug and play configuration mode. Akuvox devices will configure themselves automatically when users power on the devices and connect them to the network. It is a super time-saving mode and it will greatly bring users convenience by reducing manual operations. This mode requires no prior configurations previously by the

administrator.

2. **Cloud mode:** Akuvox Cloud is an all-in-one management system. Akuvox Cloud is a mobile service that allows audio, video, remote access control between smart phones and Akuvox intercoms. All configurations in the device will be issued automatically from cloud. If users decide to use Akuvox cloud, please contact Akuvox technical support, and they will help you configure the related settings before using it.
3. **SDMC mode:** SDMC (**SIP Device Management Controller**) is a simple and comprehensive software for building management. It provides a topography for a community while offering you a graphical configuration interface for the door access, intercom, monitoring, alarm, etc. It is a convenient tool for property managers to manage, operate and maintain the community.

## ● Tool selection

Akuvox has many configuration tools for you to set up devices more conveniently. Here we list some common tools, please contact your administrator to get the tool if you need them.

1. **SDMC:** SDMC is suitable for the management of Akuvox devices in large communities, including access control, resident information, remote device control, etc.
2. **Akuvox Upgrade tool:** Upgrade Akuvox devices in batch on a LAN (**Local Area Network**)
3. **Akuvox PC Manager:** Distribute all configuration items in batch on a LAN.
4. **IP scanner:** it is used to search Akuvox device IP addresses on a LAN.
5. **FacePro:** Manage face data in batch for the door phone on a LAN.

## 5. Access the Device

R28A series system setting can be accessed on the device web interface. And it can also be accessed on the device screen for some basic settings.

### 5.1. Obtain Device IP Address

Searching the device IP by the IP scanner in the same LAN network. Just click **Scan** tab in the IP scanner to check the device IP. Or checking the device IP address from the device setting screen. Please refer to [chapter 5.2](#) for device settings.

The screenshot shows the 'IP Scanner' interface. At the top, it says 'Online Device : 7'. Below this is a search input field with 'Search' and 'Refresh' buttons. A table lists the following data:

Index	IP Address	Mac Address	Model	Room Number	Firmware Version
1	192.168.35.102	0C11050A7F9B		1.1.1.1.1	111.30.1.216
2	192.168.35.103	0C11050BE577	R20	1.1.1.1.1	20.30.4.10
3	192.168.35.104	0C11050B00B4	R20	1.1.1.1.1	20.30.4.10
4	192.168.35.107	0C11050B083F	C317	1.1.1.1.1	117.30.2.831
5	192.168.35.101	0C11050785A9	R27	1.1.1.1.1	27.30.5.1
6	192.168.35.105	A8102020128A		1.1.1.1.1	915.30.1.15
7	192.168.35.109	0C11050A5951	R29	1.1.1.1.1	29.30.2.16



**Note:**

- Only R27A/R28A supports checking IP address from device setting screen.

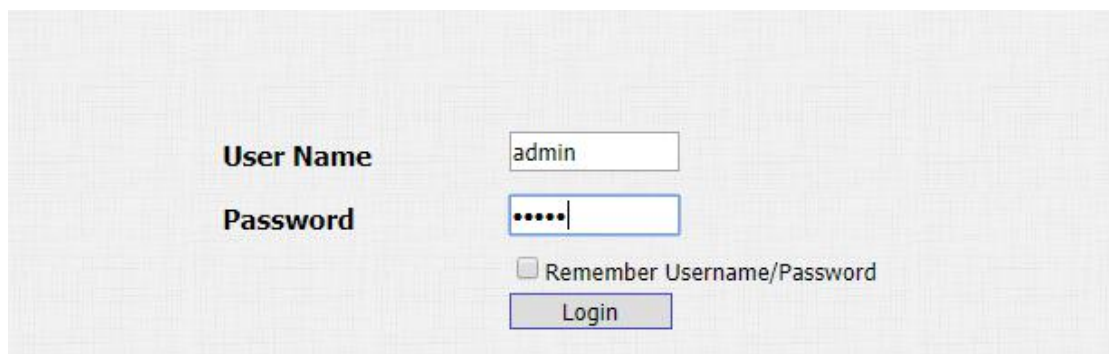
## 5.2. Access the Device Setting on the Device

### 5.2.1. Access Advance Setting Screen

Press “\*2396#” to enter the advanced setting screen. It provides some advanced permissions like editing network, reset, admin password modification to administrators, including “**System Information**,” “**Admin Settings**” and “**System Settings**”.

## 5.3. Access the Device Setting on the Web Interface

Enter the device IP address on the web browser in order to log in to the device web interface where you can configure and adjust parameters, etc. The initial user name and password are all “**admin**” and please be case-sensitive to the user names and passwords entered.



The screenshot shows a login form with the following elements:

- User Name**: A text input field containing the text "admin".
- Password**: A text input field containing five dots, indicating a masked password.
- Remember Username/Password
- Login**: A button to submit the login information.

**Note:**

- You can also obtain the device IP address using the Akuvox IP scanner to log in the device web interface. Please refer to the URL below for the IP scanner application:  
[http://wiki.akuvox.com/doku.php?id=tool:ip\\_scanner&s\[\]=ip&s\[\]=scanner](http://wiki.akuvox.com/doku.php?id=tool:ip_scanner&s[]=ip&s[]=scanner)

**Note:**

- Google Chrome browser is strongly recommended.



## 6. Language and Time Setting

### 6.1. Language Setting

When you first set up the device, you might need to set the language to your need or you can do it later if needed. And the language can be set up on the device web **Phone > Time/Lang > Web Language/LCD Language** interface according to your preference.

The screenshot shows two sections for language configuration. The first section is titled "Web Language" and contains a "Type" label followed by a dropdown menu currently set to "English". The second section is titled "LCD Language" and also contains a "Type" label followed by a dropdown menu currently set to "English".

#### Parameter Set-up:

- **Type:** choose a suitable web language. Normally, English is the default web and LCD language.

### 6.2. Time Setting

The set-up on the device web **Phone > Time/Lang > Type/NTP** interface is identical with the setting on the device, it however allows you to set up the NTP server address that you obtained to automatically synchronize your time and date. And when your time zone is selected, the device will automatically notify the NPT server of its time zone in order that the NTP server can synchronize the time zone set-up with your device.

### Type

Manual  
 Auto

Date       Year     Mon     Day  
 Time       Hour     Min     Sec

---

### NTP

Time Zone     

Primary Server     

Secondary Server     

Update Interval       (>= 3600s)

System Time      10:20:19

**Parameter Set-up:**

- **Manual/Auto:** you can choose automatically to gain the time or manually configure the date and time.
- **Date:** it is available when you choose Manual. Set up the year, month and day according to your need.
- **Time:** it is available when you choose Manual. Set up the hour, minute and second according to your need.
- **Time Zone:** it is available when you choose Auto. Select the specific time zone depending on where the device is used and then press **Confirm** tab for the confirmation. The default time zone is **GMT GMT+0.00**.
- **Primary/Secondary Server:** it is available when you choose Auto. The time zone server, normally will automatically obtain the time when connecting to the network. The secondary server will take effect when the primary server is invalid.
- **Update Interval:** it is available when you choose Auto. To configure interval between two consecutive NTP requests.

## 7. LED Setting

### 7.1. Infrared LED Setting

Infrared LED is applied in a dark environment in which a resident might not be able to see a visitor clearly via the video from the door phone. If the infrared LED is turned off, the door phone will turn to night mode so that you can have a clear view of the visitor. You can set up it on the device web **Intercom > Advanced** interface.

**Photoresistor**

---

Photoresistor Setting	<input style="width: 50px; text-align: center;" type="text" value="1500"/> - <input style="width: 50px; text-align: center;" type="text" value="1600"/> (0~1800)
Now	<input style="width: 50px; text-align: center;" type="text" value="1174"/> <input style="width: 50px; text-align: center;" type="button" value="Read"/>

#### Parameter Set-up:

- **Photoresistor Setting:** set the triggering points for disabling the infrared LED and enabling the infrared LED. For example, if you set the triggering points at 1500-1600, the infrared LED will be enabled when the photoresistor value gets higher than 1600 (the video image you see from the door phone will become black and white) and if the value gets lower than 1500, then the infrared light will be disabled ( the video image you see will have color).
  
- **Now:** click **Read** to obtain the current environment brightness and use it as a reference to set up the photoresistor setting.

## 7.2. LED Setting on Card Reader Area

You can enable or disable the LED lighting on the card reader area as needed on the web interface. Meanwhile, if you do not want to have the LED light on the card reader area stay on, you can also set the timing for the exact time span during which the LED light can be disabled in order to reduce the electrical power consumption. To do this configuration on the web **Intercom > LED Setting** interface.

Card LED Enable	Disabled	▼
Start Time (H)	18	- 06 (0~24)

### Parameter Set-up:

- **Card LED Enable:** if enabled, the LED light will stay on during the time interval (Start Time (H)). If disabled, the LED light will stay off and will turn on only when you swipe cards or press the keypad, and the time interval will become non-editable.
- **Start Time (H):** set the time interval for the light to stay on. enter the time interval for the LED light to be turned on, e.g. if the time interval is set from **8-0 (Sart time- End time)** it means LED light will stay on during the time span from **8:00 am to 12:00 pm** during one day (24 hours). This setting can not be set up when you disable the LED light.

## 7.3. LED Settings on Keypad

You can enable or disable the LED lighting of keypad as needed on the web interface. Meanwhile, If you prefer not to have the LED light of keypad stay on, you can also set the timing for the exact time span during which the LED light can be enabled in order to reduce the electrical power consumption, etc. To do this configuration on the web **Intercom > LED Setting** interface.

Start Time (H)	<input type="text" value="18"/> - <input type="text" value="06"/> (0~24)
KeyPad LED Enable	<input type="text" value="Disabled"/> ▾

### Parameter Set-up:

- **Keypad LED Enable:** Click to enable or disable the keypad LED lighting.
- **Start Time (H):** Enter the time span for the LED lighting to be valid. Eg. If the time span is from 18-22 it means LED light will stay on during the time span from 6:00 pm to 22:00 pm during a day.

## 7.4. LED Settings on Screen

You can enable or disable the LED lighting of the screen as needed on the web interface. Meanwhile, If you prefer not to have the LED light of screen stay on, you can also set the timing for the exact time span during which the LED light can be enabled in order to reduce the electrical power consumption, etc.

Wake Mode	<input type="text" value="Auto"/> ▾
Screen LED Enable	<input type="text" value="Disabled"/> ▾
Start Time (H)	<input type="text" value="18"/> - <input type="text" value="06"/> (0~24)

### Parameter Set-up:

- **Wake Mode:** There are two modes - Auto and Manual for waking up the device when idle. If you select **"Auto"** mode then the screen will be

awakened when someone approaches, the IR sensor will be triggered. And if **“Manual”** mode is selected, then you have to touch the keypad.

- **Screen LED Enable:** click to enable or disable the screen LED lighting.
- **Start Time(H):** enter the time span for the LED lighting to be valid. Eg. If the time span is from 18-22 it means LED light will stay on during the time span from 6:00 pm to 22:00 pm during a day.

## 7.5. LCD Text Display

You can customize the LCD text during the idle by themselves, such as “Welcome” or something else. Only R27X and R28A support this function, you can do this configuration on the web **Intercom > Advanced** interface.

LCD Display	
LCD Display	Default ▾
LCD Text	Press call button

### Parameter Set-up:

- **LED Display:** there are four modes - **Default**, **Hide Contacts**, **Text Only** and **Contacts Only**. If **Default** is selected, the main screen shows **Call**, **Contacts**, **PIN Entry** and **Security Center**. If **Hide Contacts** is selected, the main screen shows **Call**, **PIN Entry** and **Security Center**. If **Text Only** is selected, it will only show the customized text you entered. If **Contacts Only** is selected, it will only show **Contacts**.
- **LCD Text:** it is available when you choose **Text Only**. Enter the display content you need.

## 7.6. Backlight Setting

If you want to brighten up the screen in order to see the screen at greater ease in an environment with higher light intensity, you need to set up the related parameters in web **Intercom > LED Setting > LED Control** interface.

Backlight Value	<input type="text" value="180"/>	(0~255)
Backlight Stand by Value	<input type="text" value="0"/>	(0~255)

### Parameter Set-up:

- **Backlight Stand by Value:** adjust the backlight for the screen in the standby mode with the value ranging from 0-255.
- **Backlight value:** set the backlight value when the device is working with the value ranging from 0-255.



**Note:**

- **Only R27A and R28A support Backlight Settings.**

## 7.7. Standby Mode Setting

Standby mode is mainly a function for screen protection. You can make the device go into idle status for a predefined time span when there is no operation on the device or no one is detected approaching the device screen. To do this configuration on the web **Intercom > LED Setting > LED Control** interface.

Standby Mode	<input type="text" value="Disabled"/>	▼
Time Out	<input type="text" value="15"/>	(5~300)

**Parameter Set-up:**

- **Standby Mode:** enable the standby feature, it screen will enter sleep mode within the time out value if there is no operation.
- **Time Out:** select the time duration from 15 seconds to 180 seconds before the device goes into idle status if the screen is not awakened. For example, if you set the "**Standby Time**" as 30 seconds, then the screen will go idle (Await screen) when the screen is not awakened for 30 seconds.



## 8. Volume and Tone Configuration

Volume and tone configuration in Akuvox door phone refers to the microphone volume, speaker volume, temper alarm volume, ringback tone and open door tone configuration. Moreover, you can upload the tone you like to enrich your personalized user experience.

### 8.1. Volume Configuration

To set up the volumes, you can set up on device web **Phone > Voice** interface.

Mic Volume	
Mic Volume	<input type="text" value="8"/> (1~15)

Volume Level	
Volume Level	<input type="text" value="1"/> ▾

Speaker Volume	
Speaker Volume	<input type="text" value="15"/> (1~15)

Tamper Alarm Volume	
Tamper Alarm Volume	<input type="text" value="15"/> (1~15)

KeyPad Volume	
KeyPad Volume	<input type="text" value="8"/> (1~15)

#### Parameters Set-up:

- **Mic Volume:** adjust the mic volume as needed.
- **Volume Level:** control the volume of all speakers. The default is 1, the first level of volume, the volume range is roughly 80-95, and 2 is the

second level of volume, the volume range is roughly 95-109.

- **Speaker Volume:** adjust the speaker volume as needed.
- **Tamp Alarm Volume:** adjust the volume for the tamper alarm.
- **Keypad volume:** adjust the volume of the keypad.

### 8.1.1. Open Door Tone Configuration

You can not only enable or disable the Open Door Tone but also controls the prompt words that accompany the tone on the web **Phone > Voice > Open Door Warning** interface.



Open Door Warning	
Open Door Succ Warning	Enabled ▾
Open Door Failed Warning	Enabled ▾

#### Parameters Set-up:

- **Open Door Success Warning:** click the field **Enabled** or **Disabled** depending on if you want to hear the prompt words that accompany that **Open Door Success** tone.
- **Open Door Failed Warning:** click the field **Enabled** or **Disabled** depending on if you want to hear the prompt words that accompany that **Open Door Failed** tone.

### 8.1.2. Upload Tone Files

You can configure door phone ringback tone and other tones related to the door opening.

### 8.1.2.1. Upload Ringback Tone

You can customize the ringback tone if you need. Please follow the prompt about the file size and format. On the web, navigate to **Phone > Voice > Ringback Tone**.

**RingBack Upload**

---

No file chosen

File Format: wav, size: < 200KB, samplerate: 16000, Bits: 16

### 8.1.2.2. Upload Open Door Tone

You can customize the tone of door open successful/failed if you need. Outside tone is used to open door via card or DTMF or PIN code. Inside tone is used to open door via triggered input interface. Please follow the prompt about the file size and format. On the web, navigate to **Phone > Voice**.

**Opendoor Outside Tone Setting**

---

Opendoor Outside Tone Setting

No file chosen

File Format: wav, size: < 200KB, samplerate: 16000, Bits: 16

**Opendoor Inside Tone Setting**

---

Opendoor Inside Tone Setting

No file chosen

Broadcast Delay  (0-60Sec)

Broadcast Frequency  (1-5)

Broadcast Interval  (0-60Sec)

File Format: wav, size: < 200KB, samplerate: 16000, Bits: 16

**Opendoor Failed Tone Upload**

---

No file chosen

File Format: wav, size: < 200KB, samplerate: 16000, Bits: 16

**Parameter Set-up:**

- **Open Door Inside Tone:** warning tone that will go off when you opened the door from inside. Open door succeeded inside warning is what you can hear when you open the door by pressing the Exit button in side.
- **Broadcast Delay:** set open door success announcement delay from 0-60 seconds. For example, if you set it as "2" seconds, then you will hear the announcement 2 seconds after the successful door unlock.
- **Broadcast Frequency:** set the number of open door success announcements.
- **Broadcast Interval:** set the interval between the two open door success announcements.
- **Open Door outside Tone:** warning tone that will go off when you opened the door from outside. Open door succeeded outside warning is what you can hear when you are granted door access via access methods on the door phone.

## 9. Network Setting

### 9.1. Network Status

To check the network status on the web **Status > Network Information** interface.

Network Information	
LAN Port Type	DHCP Auto
LAN Link Status	Connected
LAN IP Address	192.168.1.3
LAN Subnet Mask	255.255.255.0
LAN Gateway	192.168.1.1
LAN DNS1	192.168.1.1
LAN DNS2	192.168.1.1

### 9.2. Device Network Configuration

You can check for the door phone's network connection info and configure the default DHCP mode (**Dynamic Host Configuration Protocol**) and static IP connection for the device on the device web **Network > Basic** interface.

**LAN Port**

---

DHCP  
 Static IP

IP Address	<input type="text" value="192.168.1.100"/>
Subnet Mask	<input type="text" value="255.255.255.0"/>
Default Gateway	<input type="text" value="192.168.1.1"/>
LAN DNS1	<input type="text" value="8.8.8.8"/>
LAN DNS2	<input type="text"/>

**Parameter Set-up:**

- **DHCP:** select the **DHCP** mode by moving the toggle switch to the right. DHCP mode is the default network connection. If the DHCP mode is turned on, then the door phone will be assigned by the DHCP server with IP address, subnet mask, default gateway and DNS server address automatically.
- **Static IP:** select the static IP mode by checking off the DHCP check box. When static IP mode is selected, then the IP address, subnet mask, default gateway, and DNS servers address have to be manually configured according to your actual network environment.
- **IP Address:** set up the IP Address if the static IP mode is selected.
- **Subnet Mask:** set up the subnet Mask according to your actual network environment.
- **Default Gateway:** set up the correct gateway default gateway according to the IP address of the default gateway.
- **LAN DNS1/2:** set up preferred or alternate DNS Server (**Domain Name Server**) according to your actual network environment. Preferred DNS server is the primary DNS server address while the alternate DNS server is the secondary server address and the door phone will connect to the alternate server when the primary DNS server is unavailable.

### 9.3. Device Deployment in Network

Door phones should be deployed before they can be properly configured in the network environment in terms of their location, operation mode, address

and extension numbers as opposed to other devices for device control and the convenience of the management. So you can do it on web **Network > Advanced > Connect Setting** interface.

**ConnectSetting**

---

Connect Type	<input type="text" value="Cloud"/>
Discovery Mode	<input type="text" value="Enabled"/>
Device Address	<input type="text" value="1"/> . <input type="text" value="1"/> . <input type="text" value="1"/> . <input type="text" value="1"/> . <input type="text" value="1"/>
Device Extension	<input type="text" value="1"/>
Device Location	<input type="text" value="R28"/>

**Parameter Set-up:**

- **Server Type:** it is automatically set up according to the actual device connection with a specific server in the network such as **SDMC** or **Cloud and None**. **None** is the default factory setting indicating the device is not in any server type, therefore you are allowed to choose Cloud, SMDC in discovery mode.
- **Discovery Mode:** click **“Enable”** to turn on the discovery mode of the device so that it can be discovered by other devices in the network, and click **“Disable”** if you want to conceal the device so as not to be discovered by other devices.
- **Device Address:** specify the device address by entering device location information from the left to the right: **Community, Unit, Stair, Floor, Room** in sequence.
- **Device extension:** enter the device extension number for the device you installed.
- **Device Location:** enter the location in which the device is installed and used.

## 9.4. Device Local RTP configuration

For the device network data transmission purpose, the device needs to be set up with a range of RTP ports (**Real-time Transport Protocol**) for establishing an exclusive range of data transmission in the network. Path: **Network >**

Advanced > Local RTP interface.

Local RTP		
Min RTP Port	<input type="text" value="11800"/>	(1024~65535)
Max RTP Port	<input type="text" value="12000"/>	(1024~65535)

**Parameter Set-up:**

- **Min RTP Port:** enter the Port value in order to establish the start point for the exclusive data transmission range.
- **Max RTP port:** enter the Port value in order to establish the end point for the exclusive data transmission range.

## 9.5. NAT Setting

NAT (**Network Address Translation**) allows hosts in an organization's private intranet to transparently connect to hosts in the public domain. There is no need for internal hosts to have registered Internet addresses. It is a way to translate the internal private network IP address into a legal network IP address technology. The NAT in the device web is limited to maintaining a connection with the remote SIP server. The principle is to send a heartbeat message to the remote SIP server at a set interval after the function is turned on. Otherwise, the server may judge that the device is offline and allocate the SIP assigned to other devices, resulting in failure to connect to it in the future. To do this configuration on web **Account > Advance > NAT** interface.

NAT		
UDP Keep Alive Messages	<input type="text" value="Enabled"/>	▼
UDP Alive Msg Interval	<input type="text" value="30"/>	(5~60s)
RPort	<input type="text" value="Enabled"/>	▼

**Parameter Set-up:**

- **UDP Keep Alive Messages:** if enabled, the device will send out the message to the SIP server so that SIP server will recognize if the device is in online status.



- **UDP Alive Msg Interval:** set the message sending time interval from 5-60 seconds, the default is 30 seconds.
- **RPort:** enable the RPort when the SIP server is in WAN (**Wide Area Network**).

## 9.6. SNMP Setting

SNMP (Simple Network Management Protocol) is an Internet-standard protocol for managing devices on IP networks. SNMP is widely used in network management system to monitor network-attached devices for conditions that may draw network administrative attention. SNMP exposes management data in the form of variables on the managed systems, which describe the system configuration. These variables can then be queried by managing applications. These variables accessible via SNMP are organized in hierarchies, which are described by Management Information Bases (MIBs). To do the configuration on the web **Network > Advanced > SNMP** interface.

SNMP	
Active	Disabled <input type="button" value="v"/>
Port	<input type="text"/> (1024~65535)
Trusted IP	<input type="text"/>

### Parameter Set-up:

- **Active:** to enable or disable SNMP feature.
- **Port:** to configure SNMP server's port.
- **Trusted IP:** to configure allowed SNMP server address. It could be an IP address or any valid URL domain name.

## 9.7. VLAN Setting

Virtual Local Area Network is a logical grouping of two or more nodes that are not necessarily on the same physical network segment but which share

the same logical IP domain. To be specific, the purpose of VLAN is to separate the layer 2 broadcast domain. Within trunk links, the tagged packet will only be sent to those ports with the same VLAN ID. This is usually achieved by switch or router. User can benefit from deployed VLAN, such as:

- \*Security: if without VLAN, all hosts will be included in unique broadcast domain. Therefore, the consequence of ARP attack will affect all end devices in the organization.
- \*Performance: The nature of network broadcast is to flood frames among the network. In certain conditions, it is unnecessary to receive the broadcast frame. To save bandwidth for high efficiency, it will be better to separate broadcast domain by deploying VLAN. To do the configuration on the web **Network > Advanced > VLAN** interface.

VLAN		
LAN Port	Active	Disabled <input type="button" value="v"/>
	VID	1 (1~4094)
	Priority	0 <input type="button" value="v"/>

**Parameter Set-up:**

- **Active:** To enable or disable VLAN feature for designated port.
- **VID:** To configure VLAN ID for designated port.
- **Priority:** To select VLAN priority for designated port.

## 9.8. TR069 Setting

TR-069 (Technical Report 069) is the document number of the technical report, defined by the Broadband Forum, that specifies the "CPE WAN management protocol" or CWMP. It defines an application layer protocol for remote management of end-user devices. As a bidirectional SOAP/HTTP-based protocol, it provides the communication between Customer-Premises Equipment (CPE) and Auto-Configuration Servers (ACS). It includes both a safe auto configuration and the control of other CPE management functions within an integrated framework. To do the configuration on the web **Network > Advanced > TR069** interface.

TR069		
ACS	Active	Disabled <input type="button" value="v"/>
	Version	1.0 <input type="button" value="v"/>
	URL	<input type="text"/>
	User Name	<input type="text"/>
Periodic Inform	Password	*****
	Active	Disabled <input type="button" value="v"/>
	Periodic Interval	1800 (3~24×3600s)
CPE	URL	<input type="text"/>
	User Name	<input type="text"/>
	Password	*****

**Parameter Set-up:**

- **Active:** to enable or disable the TR069 feature.
- **Version:** to select supported TR069 version (version 1.0 or 1.1).
- **ACS/CPE:** ACS is short for auto configuration servers as server side, and CPE is short for customer-premise equipment as client side devices.
- **URL:** to configure URL address for ACS or CPE.
- **User Name:** to configure username for ACS or CPE.
- **Password:** to configure password for ACS or CPE.
- **Periodic Inform:** to enable periodically inform.
- **Periodic Interval:** to configure interval for periodic inform.



**Note:**

- TR-069 is a technical specification entitled CPE WAN Management Protocol (CWMP). It defines an application layer protocol for remote management of end-user devices.

## 9.9. Device Web HTTP Setting

This function is used to manage whether the device website is allowed to be accessed. The door phone supports two types of remote access methods HTTP and HTTPS(encryption). To do this configuration on the web **Network > Advanced > Web Server** interface.

Web Server	
Http Enable	Enabled ▼
Https Enable	Enabled ▼
Http Port	80 (80,1024~65534)

### Parameters Set-up:

- **Http Enable:** set whether HTTP access to the device webpage is allowed, Enabled is allowed, Disabled is not allowed, the default is Enabled.
- **Https Enable:** set whether HTTPS access to the device webpage is allowed, Enabled is allowed, Disabled is not allowed, the default is Enabled.
- **Http Port:** set up the port for HTTP access method. 80 is the default port.

## 10. Intercom Call

Intercom call in the device can be configured to allow you to perform a variety of customized intercom calls such as IP call and SIP call for different application scenarios.

### 10.1. IP call & IP Call Configuration

IP call can be made directly on the intercom device by entering the IP number on the device. And you can also disable the direct IP call if you do not allow IP call to be made on the device. To do this configuration on web **Phone > Call Feature > Others** interface.

Direct IP	Enabled	▼
Direct IP AutoAnswer	Enabled	▼
Direct IP Port	5060	(1~65535)

#### Parameters Set-up:

- **Direct IP Call:** click **"Enable"** or **"Disable"** to turn the direct IP call on or off. For example, if you do not allow direct IP call to be made on the device, you can click **"Disable"** to terminate the function.
- **Direct IP AutoAnswer:** click **"Enable"** or **"Disable"** to turn the direct IP call on or off when the phone automatically answer the incoming call.
- **Direct IP port :** set up the IP direct call port, 5060 is the default port.

### 10.2. SIP Call & SIP Call Configuration

You can make SIP call ( **Session Initiation Protocol** ) in the same way as you do for making the IP calls on the device. However, SIP call parameters related to its account, server, and transport type need to be configured first before you can make calls on the device.

## 10.2.1. SIP Account Registration

Akuvox door phones support two SIP accounts that can all be registered according to your applications. You can, for example, switch between them if any one of the account failed and become invalid. The SIP account can be configured on the device interface.

### 10.2.1.1. Configure SIP Account Configuration

To perform the SIP account setting on the Web **Account > Basic > SIP Account** Interface.

SIP Account	
Status	UnRegistered
Account	Account 1 ▾
Account Active	Disabled ▾
Display Label	<input type="text"/>
Display Name	<input type="text"/>
Register Name	<input type="text"/>
User Name	<input type="text"/>
Password	<input type="password"/>

#### Parameter Set-up:

- **Status:** check to see if the SIP account is registered or not.
- **Account Active:** click **Enable** or **Disable** to activate or deactivate the registered SIP account.
- **Display Name:** configure the name, for example, the device's name to be shown on the device being called to.
- **User Name:** enter the user name obtained from SIP account administrator.
- **Account:** select the exact account (Account 1&2) to be configured.

- **Display Label:** configure the device label to be shown on the device screen.
- **Register Name:** enter the SIP account register Name obtained from the SIP account administrator.
- **Password:** enter the password obtained from the SIP account administrator.

## 10.2.2. SIP Server Configuration

SIP servers can be set up for devices in order to achieve call sessions through SIP server between intercom devices. To do this configuration also on web **Account > Basic > SIP Server** interface.

SIP Server 1		
Server IP	<input type="text" value="192.168.35.11"/>	Port <input type="text" value="5060"/>
Registration Period	<input type="text" value="1800"/>	(30~65535s)

SIP Server 2		
Server IP	<input type="text"/>	Port <input type="text" value="5060"/>
Registration Period	<input type="text" value="1800"/>	(30~65535s)

### Parameter Set-up:

- **Server IP:** enter the primary server IP address number or its URL.
- **Server IP:** enter the backup SIP server IP address or its URL.
- **Port:** set up SIP server port for data transmission.
- **Registration Period:** set up SIP account registration time pan. SIP re-registration will start automatically if the account registration fails during the registration time span. The default registration period is "1800", ranging from 30-65535s.

### 10.2.3. Configure Outbound Proxy Server

An outbound proxy server is used to receive all initiating request messages and route them to the designated SIP server in order to establish call session via port-based data transmission. To set it up on the device web **Account > Basic > Outbound Proxy Server** Interface.

**Parameter Set-up:**

- **Enable Outbound:** click “Enable” and “Disable” to turn on or turn off the outbound proxy server.
- **Preferred Server IP:** enter the SIP address of the primary outbound proxy server.
- **Port:** enter the Port number to establish call session via the primary outbound proxy server
- **Alternate Server IP:** set up Backup Server IP for the backup outbound proxy server.
- **Port:** enter the port number for establishing call session via the backup outbound proxy server.

### 10.2.4. Configure Data Transmission Type

SIP messages can be transmitted in three data transmission protocols: **UDP (User Datagram Protocol)**, **TCP(Transmission Control Protocol)**, **TLS (Transport Layer Security)** and **DNS-SRV**. In the meantime, you can also identify the server from which the data come. To do this configuration on web **Account > Basic > Transport Type** interface.



**Transport Type**

---

Transport Type UDP

**Parameter Set-up:**

- **UDP:** select “UDP” for unreliable but very efficient transport layer protocol. UDP is the default transport protocol.
- **TCP:** select “TCP” for Reliable but less-efficient transport layer protocol.
- **TLS:** select “TLS” for Secured and Reliable transport layer protocol.
- **DNS-SRV:** select “DNS-SRV” to obtain DNS record for specifying the location of servers. And **SRV** not only records the server address but also the server port. Moreover, SRV can also be used to configure the priority and the weight of the server address.

## 10.3. Configure Calling Feature

You can select the SIP account for making SIP calls. On the web, navigate to **Intercom > Basic**.

**Intercom-Basic**

**Basic**

---

Select Account Auto

**Parameter Set-up:**

- **Select Account:** if you select “Auto” when both account 1 and account 2 are registered, then the call will be made from account 1, however, if either account 1 or account 2 is registered, then the call will only be made from the registered account. You cannot make any calls from the account1/2 that are not registered.

### 10.3.1. DND

DND ( **Do not disturb**) setting allows you not to be disturbed by any unwanted incoming SIP calls. You can set up DND related parameters properly on the device web **Phone > Call Feature > DND** interface to block SIP calls you do not intend to answer. In the meantime, you can also define the code to be sent to the SIP server when you want to reject the call.

**Parameter Set-up:**

- **Account:** select **"Account1"**, **"Account2"** or **"All account"** for the DND application.
- **DND:** enable or disable the DND function. DND function is disabled by default.
- **Return Code When DND:** select what code should be sent to the calling device via SIP server. **404** for **"Not found"**; **480** for **"Temporary unavailable"** **486** for **"busy here"**.
- **DND On Code:** turn on the DND on server using the Code obtained. The DND on Code is **78** by default.
- **DND Off Code:** turn off the DND on server using the code obtained. The DND off Code is **79** by default.
- **Return Code When Refuse:** select code to be sent to the caller side via SIP server when you rejected the incoming call.

**10.3.2. Speed Dial Call**

Speed Dial is used to quickly initiate the pre-configured numbers by pressing **Dial** key. You can create up to 40 numbers on R28A and up to 80 numbers on R27A. To do the configuration on the web **Intercom > Basic > Speed Dial**

interface.

Key	Number	Number	Number	Number
Speed Dial 1	<input type="text"/>	<input type="text"/>	<input type="text"/>	<input type="text"/>
Speed Dial 2	<input type="text"/>	<input type="text"/>	<input type="text"/>	<input type="text"/>

**Parameters Set-up :**

- **Add** : click to show more numbers.

### 10.3.3. Robin Call

Robin call is used to initiate multiple numbers from one family in Akuvox SmartPlus one by one. If the previous callee does not answer within the robin call timeout, the call will be transferred to the next one. If the call is answered by one of the callees, the call will not be transferred anymore. This feature is only available when connecting Akuvox SmartPlus. To do the configuration on the web **Intercom > Basic > Robin Call** interface.

**Robin Call**

Robin Call Enable

Robin Call Timeout

**Parameters Set-up:**

- **Robin Call Enable:** it is disabled by default. It needs to be controlled by Akuvox SmartPlus.
- **Robin Call Timeout:** call out time value for each number, ranging from 5 - 60s.

### 10.3.4. Web Call

In addition to making IP/SIP call directly on the device, you can also make the call on the device web interface without approaching to device physically for testing purpose, etc. To do the configuration on the web **Intercom > Basic > Web Call** interface.



#### Parameters Set-up:

- **Auto/Account1/Account2:** to choose a suitable SIP account to make a web call. If you call out using IP address, Account selection is not need to be chosen.

### 10.3.5. Dial Option

If you want to replace the long and complex dial number with a shorter number that can be memorized at greater ease and convenience for making calls. You can replace multiple device dial numbers such as IP addresses with only one short number. To configure the number replacement on the device, navigate to **Phone > Dial Plan**, then click **Add**. **To replace the number in batch, you can import the .xml file to the door phone.** And the file from the door phone can be exported out before importing them to other door phones.

**Dial Plan**

**Rules Management**

No file chosen      (.XML)       

Index	Account	Name	Prefix	Replace 1	Replace 2	Replace 3	Replace 4	Replace 5	<input type="checkbox"/>
1	Auto	123	1	akuvox002					<input type="checkbox"/>
2									<input type="checkbox"/>
3									<input type="checkbox"/>
4									<input type="checkbox"/>
5									<input type="checkbox"/>
6									<input type="checkbox"/>
7									<input type="checkbox"/>
8									<input type="checkbox"/>
9									<input type="checkbox"/>
10									<input type="checkbox"/>

Page:               

**Rules Modify >>**

Account	Auto <input type="button" value="v"/>
Name	<input type="text"/>
Prefix	<input type="text"/>
Replace 1	<input type="text"/>
Replace 2	<input type="text"/>
Replace 3	<input type="text"/>
Replace 4	<input type="text"/>
Replace 5	<input type="text"/>

### 10.3.6. Auto Answer

You can define how quickly the door phone should respond in answering the incoming SIP/IP call automatically by setting up the time related parameters. In addition, you can also define the mode in which the calls are to be answered ( video mode or audio mode). To enable this feature on web **Account > Advanced > Call** interface, you can set up the related parameters on web **Phone > Call Feature>Others**.

Auto Answer	Enabled ▾
Auto Answer Delay	0 (0~5s)
Auto Answer Mode	Video ▾

**Parameters Set-up:**

- **Auto Answer:** Turn on the Auto Answer function by clicking "**Enable**".
- **Auto Answer Delay:** Set up the delay time (from 0-5 sec.) before the call can be answered automatically. For example, if you set the delay time as 1 second, then the call will be answered in 1 second automatically.
- **Auto Answer Mode:** Set up the video or audio mode you preferred for answering the call automatically.

### 10.3.7. Multicast

Multicast uses one-to-many mode to communicate in a range. Door phone can be a listener and receive the audio from the listened part. To do the configuration on the web **Phone > Multicast** interface.

### Multicast Setting

Paging Barge

Paging Priority Active

---

### Priority List

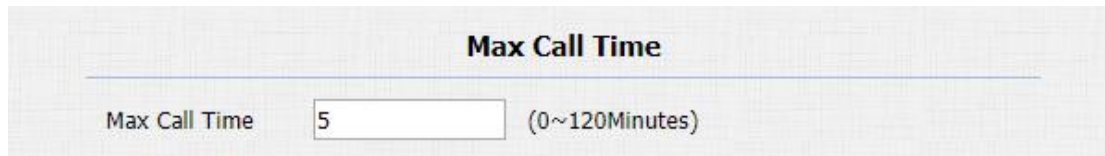
IP Address	Listening Address	Label	Priority
1 IP Address	<input type="text" value="224.1.6.11:1200"/>	<input type="text" value="Akuvox"/>	1
2 IP Address	<input type="text"/>	<input type="text"/>	2
3 IP Address	<input type="text"/>	<input type="text"/>	3
4 IP Address	<input type="text"/>	<input type="text"/>	4
5 IP Address	<input type="text"/>	<input type="text"/>	5
6 IP Address	<input type="text"/>	<input type="text"/>	6
7 IP Address	<input type="text"/>	<input type="text"/>	7
8 IP Address	<input type="text"/>	<input type="text"/>	8
9 IP Address	<input type="text"/>	<input type="text"/>	9
10 IP Address	<input type="text"/>	<input type="text"/>	10

**Parameters Set-up:**

- **Paging Barge:** multicast or how many multicast calls are higher priority than SIP call, if you disable Paging Priority Active, SIP call will have high priority.
- **Paging Priority Active:** multicast calls are called in order of priority or not.
- **Listening Address:** enter the multicast IP address you want to listen. The multicast IP address needs to be the same as the listened part and the multicast port can not be the same for each IP address. Multicast IP address is from 224.0.0.0 to 239.255.255.255.
- **Label:** enter the label for each listening address.

### 10.3.8. Configure Maximum Call Duration

Door phone allows you to set up the call time duration in receiving the call from the calling device as the caller side might forget to hang up the intercom device. When the call time duration is reached, the door phone will terminate the calling automatically. To do this configuration on the web **Intercom > Basic > Max Call Time** interface.



#### Parameters Set-up:

- **Max Call Time:** enter the call time duration according to your need (Ranging from 0-120 min.). The default call time duration is 5 min.



#### Note:

- Max call time of the device is also related to max call time of the SIP server. If using SIP account to make a call, please pay attention to the max call time of the SIP server. If the max call time of SIP server is shorter than the max call time of the device, the shorter one is available.

### 10.3.9. Maximum Dial Duration

Maximum Dial duration consists of Maximum dial in time duration and the maximum dial out time. Maximum dial in time refers to the maximum time duration before the door phone hangs up the call if the call is not answered by the door phone. On contrary, Maximum dial out time refers to the maximum time duration before the door phone hangs up itself automatically when the call from the door phone is not answered by the intercom device being called. To do this configuration on the web **Intercom > Basic > Max Dial Time** interface.



**Max Dial Time**


---

Dial In Time  (1~120Sec)

Dial Out Time  (1~120Sec)

**Parameters Set-up:**

- **Dial in Time:** enter the dial in time duration for your door phone (ranging from 30-120 sec). For example, if you set the dial in time duration is 60 seconds in your door phone, then the door phone will hang up the incoming call automatically if the call is not answered by the door phone in 60 seconds. 60 seconds is the dial in time duration by default.
- **Dial out Time:** enter the dial in time duration for your door phone (ranging from 5-120 sec). For example, if you set the dial out time duration is 60 seconds in your door phone, then the door phone will hang out the call it dialed out automatically if the call is not answered by the device being called.

 **Note:**

- Max call time of the device is also related to max call time of the SIP server. If using SIP account to make a call, please pay attention to the max call time of the SIP server. If the max call time of SIP server is shorter than the max call time of the device, the shorter one is available.

### 10.3.10. Hang Up After Open Door

This feature is used to hang up the call automatically after the door is released during a call. So the caller or callee does not need to click hang up key again. To do this configuration on the web **Intercom > Basic > Hang Up After Open Door interface.**

**Hang Up After Open Door**

---

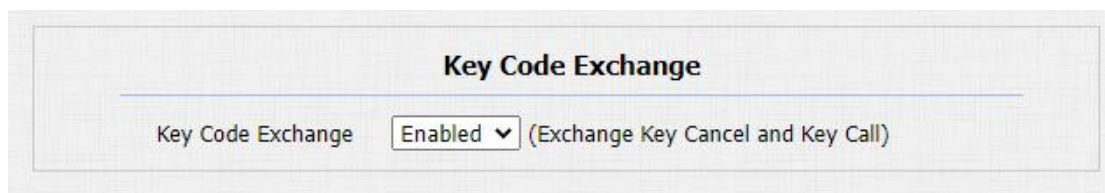
Timeout  (0~15)

**Parameter Set-up :**

- **Timeout:** the time out value can be set up from 1 second to 15seconds. 5 seconds is the default. The call will be automatically hang up within this value after the door is opened.

### 10.3.11. Switch Cancel&Dial Key

On R28 door phones, the **Cancel** Keys and **Dial** keys can be different in their positions functionally and physically. Either the Cancel key is right on the top of the Dial key or the other way round. You can reverse their functional positions on the web to match their physical position on the keypad. On the web, navigate to **Intercom > Advanced > Key Code Exchange**.

**Parameter set-up:**

- **Key Code Exchange:** if enabled, the **Cancel** function will be placed right on the top of Dial function in their functional location. If disabled, their functional location will be functionally reversed.

# 11. Audio & Video Codec Configuration

## 11.1. Audio Codec Configuration

Akuvox door phone supports four types of Codec (PCMU, PCMA, G729, G722) for encoding and decoding the audio data during the call session. Each type of Codec varies in terms of sound quality. You can select the specific codec with different bandwidths and sample rates flexibly according to the actual network environment. It is only available for SIP call. To do the configuration on device web **Account > Advanced** interface.

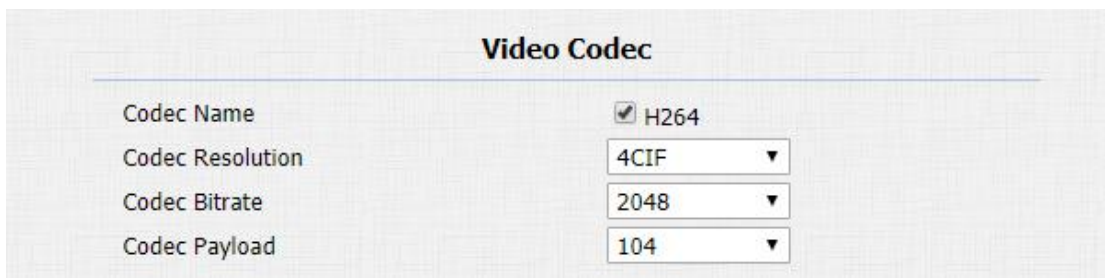
The screenshot shows the 'SIP Account' configuration page. At the top, there is a dropdown menu for 'Account' set to 'Account 1'. Below this is the 'Codecs' section, which is divided into two columns: 'Disabled Codecs' and 'Enabled Codecs'. The 'Enabled Codecs' list contains four items: PCMU, PCMA, G722, and G729. Between the two lists are two buttons: '>>' and '<<'. To the right of the 'Enabled Codecs' list are two buttons: an upward arrow and a downward arrow.

Please refer to the bandwidth consumption and sample rate for the four codecs types below:

Codec Type	Bandwidth Consumption	Sample Rate
PCMA	64 kbit/s	8kHz
PCMU	64 kbit/s	8kHz
G729	8 kbit/s	8kHz
G722	64 kbit/s	16kHz

## 11.2.Video Codec Configuration

Akuvox door phone support H264 codec that provides a better video quality at a much lower bit rate with different video quality and payload. It is only available for SIP call. To set up video codec on web **Account > Advanced** interface.



The screenshot shows a configuration window titled "Video Codec". It contains four settings, each with a label and a value:
 

- Codec Name:** H264 (with a checked checkbox)
- Codec Resolution:** 4CIF (dropdown menu)
- Codec Bitrate:** 2048 (dropdown menu)
- Codec Payload:** 104 (dropdown menu)

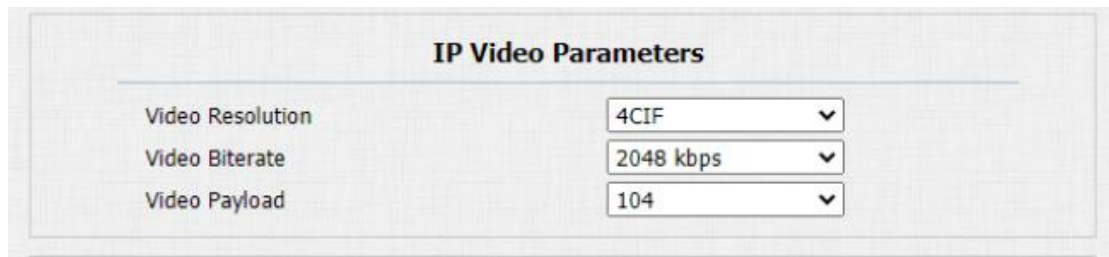
### Parameter Set-up:

- **Codec Name:** check to select the H264 video codec format for the door phone video stream. H264 is the video codec by default.
- **Codec Resolution:** select the code resolution for the video quality among four options: "QCIF", "CIF", "VGA", "4CIF" and "720P" according to your actual network environment. The default code resolution is 4CIF.
- **Codec Bitrate:** select the video stream bit rate (Ranging from 320-2048). The greater the bitrate, the data transmitted every second is greater in amount therefore the video will be clearer. While the default code bitrate is 2048.
- **Codec Payload:** select the payload type (ranging from 90-118) to

configure the audio/video configuration file. The default payload is 104.

## 11.3. Video Codec Configuration for IP Direct Calls

You can select the IP call video quality by selecting the proper codec resolution according to your actual network condition. To do so, you can go to **Phone>Call Feature > IP Video Parameters**.



IP Video Parameters	
Video Resolution	4CIF
Video Biterate	2048 kbps
Video Payload	104

### Parameter Set-up:

- **Video Resolution:** select the code resolution for the video quality among four options: "CIF", "VGA", "4CIF" and "720P". The default code resolution is 4CIF.
- **Video Bitrate:** select video bit-rate among six options: "64 kbps", "256kbps", "512 kbps", "1024 kbps", "2048 kbps" according to your network environment. The default video bit-rate is "2048 kbps".
- **Video Payload:** select the payload type (ranging from 90-118) to configure the audio/video configuration file. The default payload is 104.

## 11.4. Configure DTMF Data Transmission

In order to achieve the door access via DTMF code or some other applications, you are required to properly configure DTMF on web **Account > Advanced > DTMF** in order to establish a DTMF-based data transmission between the door phone and other intercom devices for the third party integration.

DTMF	
Type	RFC2833 <input type="button" value="v"/>
How To Notify DTMF	Disabled <input type="button" value="v"/>
DTMF Payload	101 (96~127)

**Parameter Set-up:**

- **Type:** select DTMF mode among five options: **“Inband”**, **“RFC2833”**, **“Info+Inband”** and **“Info+RFC2833”** based on the specific DTMF transmission type of the third party device to be matched with as the party for receiving signal data.
- **How to Notify DTMF:** select among four types: **“Disable”** **“DTMF”** **“DTMF-Relay”** **“Telephone-Event”** according to the specific type adopted by the third party device. You are required to set it up only when the third party device to be matched with adopts **“Info”** mode
- **DTMF Payload:** set the payload according to the specific data transmission payload agreed on between the sender and receiver during the data transmission.

## 12. Phone Book Configuration

Akuvox door phone supports storing up to 500 contacts that can give access permission to the indoor monitor or another device. Access White list includes group setting and contact setting and management. To set it up on web **PhoneBook > Local Book** interface.

### 12.1. Managing Contact Group

You can set up contact group first before you move the contacts into the group, to do so navigate to on web **PhoneBook > Local Book > Group**.

**Group**

Index	Name	Firstly Called	Secondary Called	Lastly Called	
1	akuvox	inn,Ann	Lin	Julia	<input type="checkbox"/>
2					<input type="checkbox"/>
3					<input type="checkbox"/>
4					<input type="checkbox"/>
5					<input type="checkbox"/>
6					<input type="checkbox"/>
7					<input type="checkbox"/>
8					<input type="checkbox"/>
9					<input type="checkbox"/>
10					<input type="checkbox"/>

Page: 1 ▾
Prev
Next
Delete
Delete All

**Group Setting**

Name

Add
Edit
Cancel

**Parameter Set-up:**

- **Name:** enter the group name.

## 12.2. Managing Contacts

You can search, create, display, edit and delete the contacts in your phone book. Path: **PhoneBook > Local Book**.

**Local Book**

**Contact**  ▼

**Contacts Sort By**  ▼

**Show Cloud Contacts**  ▼

**Caller Display**  ▼

**Search**

**Dial**   ▼

Index	Name	Phone Number	Group	Account	Priority Of Call	Floor	<input type="checkbox"/>
1							<input type="checkbox"/>
2							<input type="checkbox"/>
3							<input type="checkbox"/>
4							<input type="checkbox"/>
5							<input type="checkbox"/>
6							<input type="checkbox"/>
7							<input type="checkbox"/>
8							<input type="checkbox"/>
9							<input type="checkbox"/>
10							<input type="checkbox"/>

Page:  ▼

**Contact Setting**

Name	<input type="text"/>	Phone Number	<input type="text"/>
Group	<input type="text" value="Default"/> <span style="float: right;">▼</span>	Account	<input type="text" value="Auto"/> <span style="float: right;">▼</span>
Priority Of Call	<input type="text" value="Firstly Called"/> <span style="float: right;">▼</span>	Floor	<input type="text" value="0"/> <span style="float: right;">▼</span>

### Parameters Set-up:

- **Contact:** you can choose to show all contacts information or one group's contact information.
- **Contacts short by:** there are three options **ASCII Code**, **Room Number** and **Import**. If **ASCII Code** is selected, sort in ascending ASCII order, for example: 0-9, a-z, numbers take precedence over letters. Not case



sensitive, but the same letter, lowercase is sorted before uppercase. If **Room Number** is selected, sort by room name. if there is no room name, the room number is taken as the room name by default. Room number is available after enabling Cloud contact. If **Import** is selected, sort by contacts in the imported file

- **Show Cloud Contacts:** enable it to show the contacts issued from Akuvox SmartPlus.
- **Caller display:** display the caller by name or group name.
- **Search:** enter the key number or key letter of the name to quickly search the contact.
- **Dial:** enter a phone number then click Dial to initiate the call from the web.
- **Name:** enter the contact name, which is required.
- **Phone:** enter the phone number of the contact, which is required.
- **Group:** click the green tab to select the group name you have created. You cannot select the group name if no group name has been created.
- **Account:** select which SIP account will be used to call out. If using IP direct call, it is not available.
- **Priority of Call:** up to 3 numbers in one group and set up the call sequence for these numbers.
- **Floor:** enter the floor number of the contact if needed.

### 12.3. Export/Import Contacts

When the contact becomes so many that you can not afford to manage each contact one by one manually, you can import and export the contacts in batch on the device web interface.

**Import/Export**

---

**Contact**  No file chosen (.XML)

# 13. Relay Setting

## 13.1. Relay Switch Setting

You can configure the relay switch(es) and DTMF for the door access on the web **Intercom > Relay** interface.

The screenshot shows a web interface titled "Relay" with a sub-header "Relay". It contains a table of configuration options for three relays: RelayA, RelayB, and RelayC. The settings are as follows:

Parameter	RelayA	RelayB	RelayC
Relay ID	RelayA	RelayB	RelayC
Relay Type	Default state	Default state	Default state
Relay Mode	Monostable	Monostable	Monostable
Relay Delay(Sec)	3	3	3
DTMF Option	1 Digit DTMF		
DTMF	0	0	0
Multiple DTMF			
Relay Status	RelayA: Low	RelayB: Low	RelayC: Low
Opendoor Outside Tone	Default	Default	Default
Opendoor Inside Tone	Default	Default	Default

### Parameter Set-up:

- **Relay ID:** you are allowed to set up three relay switches in total for the door access control.
- **Relay Type:** if the Default state is selected, the Relay Status shows Low which means the door is closed, the Relay Status shows High which means the door is opened. If Invert State is selected, the Relay Status shows High which means the door is closed, and Low means the door is opened.
- **Relay Mode:** there are two modes Monostable and Bistable. If Monostable is selected, the relay status will be automatically reset within the relay delay time after the relay is triggered. If Bistable is selected, relay status will be reset after the relay is triggered again.

- **Relay Delay (Sec):** set the relay trigger delay timing (Ranging from 1-10 Sec.) For example, if you set the delay time as "5" Sec. Then the relay will not be triggered until 5 seconds after you press "**unlock**" tab.
- **DTMF Option:** select the number of DTMF digits for the door access control (**Ranging from 1-4 digits** ). For example, you can select 1 digit DTMF code or 2-digit DTMF code, etc, according to your need.
- **DTMF:** set the 1-digit DTMF code within range from ( **0-9 and \*,#**).
- **Multiple DTMF:** set the DTMF code according to the **DMTP Option** setting. For example, you are required to set the 3-digits DTMF code if **DTMF Mode** is set as 3-digits.
- **Relay Status:** relay status is low by default which means normally closed(NC) If the relay status is high, then it is in Normally Open status(NO).
- **Opendoor Outside/Inside Tone:** you can choose the customized open door tone. To configure the tone please refer to [chapter 8.1.3.2](#).

**Note:**

- Only the external devices connected to the relay switch need to be powered by power adapters as the relay switch does not supply power.

**Note:**

- If DTMF mode is set as "**1 Digit DTMF**", you cannot edit DTMF code in **2~4 Digits DTMF** field. And if you set DTMF mode from 2-4 in **2~4 Digits DTMF**" field, you can not edit DTMF code in **1 Digit DTMF** field.

## 13.2. Select Speed Dial triggered Relay

This function is used to trigger a relay when you initiate the speed dial numbers. To do this configuration on the web **Intercom > Basic > Trigger Relay Of Speed Dial** interface.



Trigger Relay Of Speed Dial

RelayID      RelayA     RelayB     RelayC

### Parameters Set-up:

- **RelayID:** tick the checkbox of the relay. You can choose one or all relays.

## 13.3. Web Relay Setting

In addition to the relay that is connected to the door phone, you can also control the door access using the network-based web relay on the device and on the device web interface. Web relay needs to be set up on the web **Phone > WebRelay** interface where you are required to fill in such information as relay IP address, password, web relay action, etc before you can achieve the door access via web relay.

**Web Relay**

Type

IP Address

UserName

Password

**Web Relay Action Setting**

Action ID	Web Relay Action	Web Relay Key	Web Relay Extension
Action ID 01	<input type="text"/>	<input type="text"/>	<input type="text"/>
Action ID 02	<input type="text"/>	<input type="text"/>	<input type="text"/>
Action ID 03	<input type="text"/>	<input type="text"/>	<input type="text"/>
Action ID 04	<input type="text"/>	<input type="text"/>	<input type="text"/>
Action ID 05	<input type="text"/>	<input type="text"/>	<input type="text"/>
Action ID 06	<input type="text"/>	<input type="text"/>	<input type="text"/>
Action ID 07	<input type="text"/>	<input type="text"/>	<input type="text"/>
Action ID 08	<input type="text"/>	<input type="text"/>	<input type="text"/>
Action ID 09	<input type="text"/>	<input type="text"/>	<input type="text"/>
Action ID 10	<input type="text"/>	<input type="text"/>	<input type="text"/>

**Parameter Set-up:**

- **Type:** select among three options **"Disabled"** **"WebRelay"** and **"Both"**. Select **"Webrelay"** to enable the web relay. Select **"Disable"** to disable the web relay. Select **"Both"** to enable both local relay and web relay.
- **IP Address:** enter the web relay IP address provided by the web relay manufacturer.
- **User Name:** enter the User name provided by the web relay manufacturer.
- **Password:** enter the password provided by the web relay manufacturer. The passwords are authenticated via HTTP and you can define the passwords using **"http get"** in Action.
- **Web Relay Action:** enter the specific web relay action command provided by the web manufacturer for different actions by the web relay. Without adding ip, username, pwd, you can fill in the HTTP command in the web relay action, so you can configure multiple webrelays.
- **Web Relay Key:** it can be null or enter the configured DTMF code, when the door is unlocked via DTMF code, the action command will be sent to the web relay automatically.

- **Web Relay Extension:** it can be null or enter the relay extension information, which can be a SIP Account user name of an intercom device such as an indoor monitor, so that the specific action command will be sent when unlock is performed on the intercom device, while this setting is optional. And please refer to the example below:  
<http://admin:admin@192.168.1.2/state.xml?relayState=2>.

## 13.4. Configure White List for Door Relay

For security, Akuvox door phone give permission for who can unlock the door by DTMF code. To do this configuration on the web **Intercom > Relay > Open Relay Via DTMF** interface.

The screenshot shows the 'Open Relay Via DTMF' configuration page. It features a form with a label 'Access Phone Numbers' and a dropdown menu for 'WhiteList Numbers'. The dropdown menu is open, showing three options: 'Disabled', 'WhiteList Numbers' (which is highlighted in blue), and 'All Numbers'. Below the dropdown menu are two buttons: 'Submit' and 'Cancel'.

### Parameter Set-up:

- **Access Phone Numbers:** there are three options - **Disabled**, **WhiteList Number** and **All Number**. If **Disabled** is selected, the DTMF code for unlock can not be available during the call. If **WhiteList** is selected, Akuvox door phone can only be unlocked by the contacts existing in the phone book. If **All Number** is selected, all callees can open the door phone during the call by using DTMF code.

## 14. Door Access Schedule Management

You are required to configure and make schedule for the user-based door access via RF card, Private PIN and Facial recognition.

### 14.1. Configure Door Access Schedule

You can create door access schedules so that they can be later conveniently applied to the door access control intended for individual users or a group of users created. Moreover, you can edit your door access schedule if needed.

#### 14.1.1. Manage Relay Schedule

Set the corresponding relay always open at a specific time. This feature is designed for some specific scenarios, for example, the time after school, or for morning work time. To do the configuration on the web **Intercom > Relay > Relay Schedule** interface.

The screenshot shows the 'Relay Schedule' configuration page. At the top, there is a title 'Relay Schedule'. Below the title, there are two dropdown menus: 'Relay ID' with 'RelayA' selected, and 'Schedule Enable' with 'Enabled' selected. Below these are two empty list boxes: 'All Schedules' on the left and 'Enable Schedules' on the right. Between these two list boxes are two buttons: '>>' and '<<'. The interface is set against a light gray grid background.

#### Parameter Set-up:

- **Relay ID:** choose the relay you need to set up.
- **Schedule Enable:** it is disabled by default. Only choose to enable it, that



you can select the schedule. For creating the schedule please refer to [chapter 14.1.2.](#)

## 14.1.2. Create Door Access Schedule

You can create the door access schedule on a daily, weekly, and monthly basis, and you can also create a schedule that allows you to plan for a longer period of time in addition to running the door access schedule on a daily or monthly basis. To do this configuration on web **Intercom > Schedules** interface.

To create a daily schedule, you can do as follows:

### Schedule Setting

Schedule Type:

Schedule Name:

Date Range:  -

Day of Week: Mon  Tue  Wed  Thur   
 Fri  Sat  Sun  Check All

Date Time:  :  -  :

### Schedules Management

Index	Schedule ID	Source	Mode	Name	Date	Day of Week	Time	
1	1002	Local	Daily	Never	-	-	-	<input type="checkbox"/>
2	1001	Local	Daily	Always	-	-	00:00:00-23:59:59	<input type="checkbox"/>
3								<input type="checkbox"/>
4								<input type="checkbox"/>
5								<input type="checkbox"/>
6								<input type="checkbox"/>
7								<input type="checkbox"/>
8								<input type="checkbox"/>
9								<input type="checkbox"/>
10								<input type="checkbox"/>

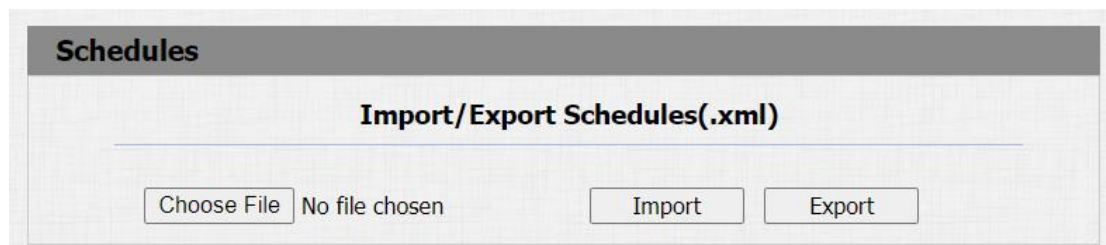
Page:

**Parameters Set-up:**

- **Schedule Type:** set the type of time period. There are three types to choose from: Daily, Weekly, and Normal. The default is Daily.
- **Schedule Name:** set the name of the time period.
- **Date Range:** set the corresponding time period.
- **Day of Week:** select the corresponding day of the week. This field will only be displayed when the Week and Normal types are selected.
- **Date Range:** set the corresponding date. This field will only be displayed when the Normal type is selected.

### 14.1.3. Import and Export Door Access Schedule

In addition to creating a door access schedule separately, you can also conveniently import or export the schedules in order to maximize your door access schedule management efficiency. Path: **Intercom > Schedule > Import/Export Schedule(.xml)**.

**Note:**

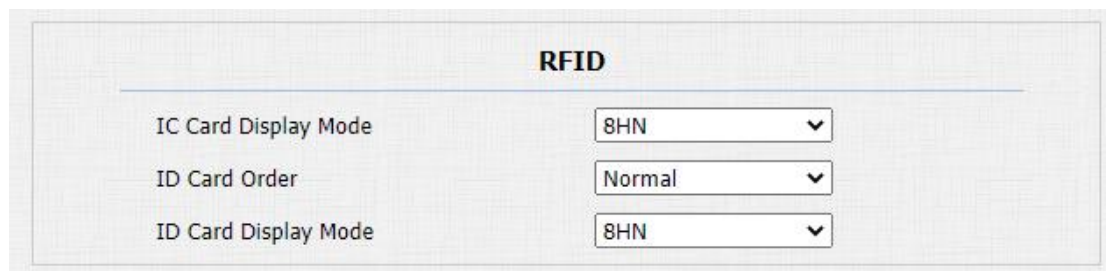
- It only supports .xml format file for importing and exporting the schedule.

## 14.2. Door Unlock Configuration

R28 door phone offers you two types of access methods: PIN code and RF card.

### 14.3. Configure Access Card Format

If you want to integrate with the third party intercom system in terms of RF card door access, you can change the RF card code format to be identical with that applied in the third party system. You can do this configuration on web **Intercom > Card Setting** interface.



The screenshot shows a configuration window titled "RFID". It contains three rows of settings, each with a label and a dropdown menu:

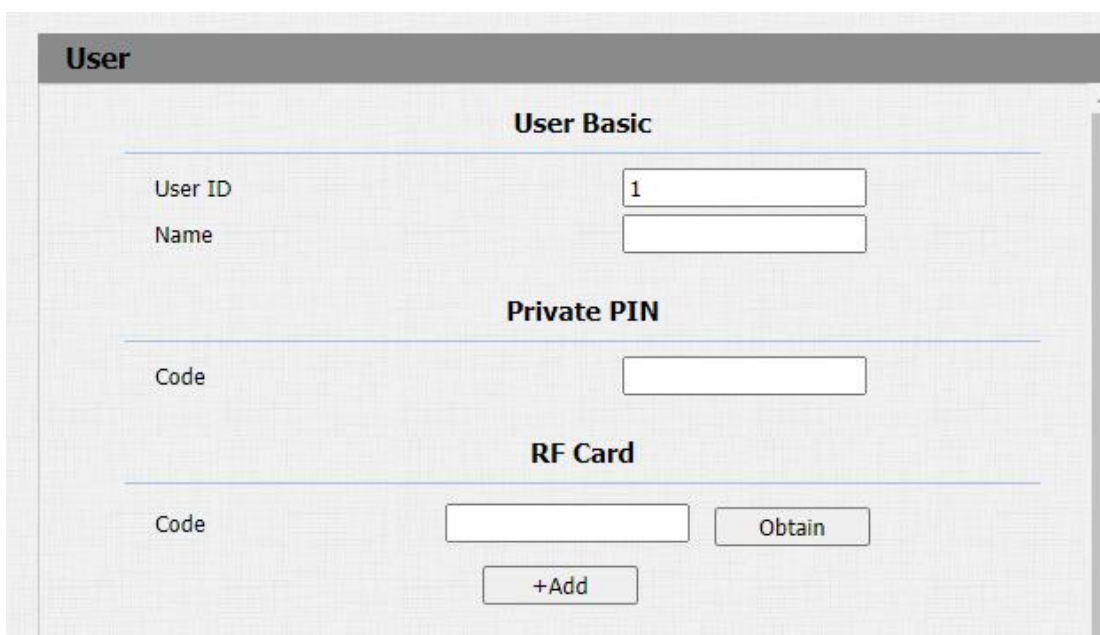
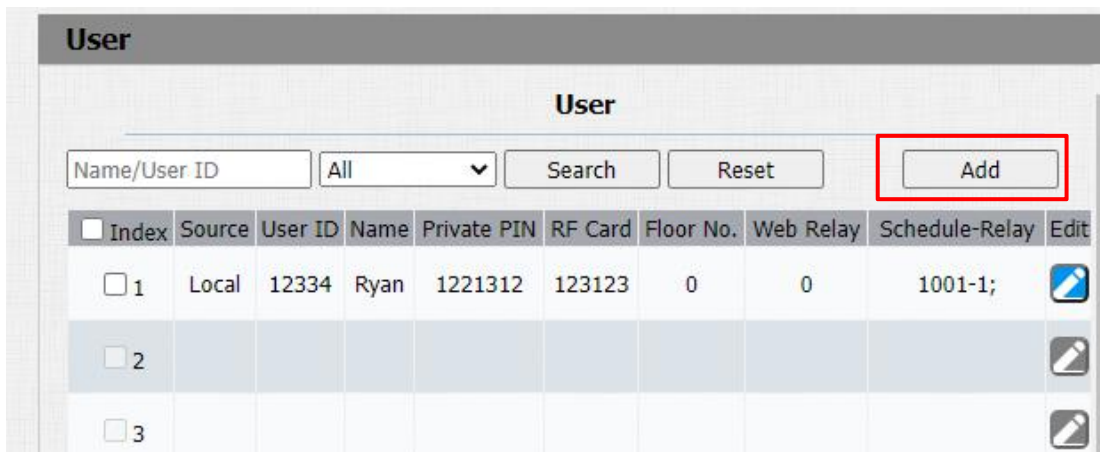
RFID	
IC Card Display Mode	8HN
ID Card Order	Normal
ID Card Display Mode	8HN

#### Parameters Set-up:

- **IC CARD Display Mode:** Select the card code format for the **IC card** for the door access among five format options: **8H10D; 6H3D5D(W26); 6H8D; 8HN; 8HR**. The card code format is 8HN by default in the door phone.
- **ID Card Order:** select normal or reversed display of ID card.
- **ID Card Display Mode:** Select the card format for the **ID Card** for the door access among five format options: **8H10D; 6H3D5D(W26); 6H8D; 8HN; 8HR**. The card code format is 8HN by default in the door phone.

## 14.4. Configure RF Card for Door Unlock

You can tap the RF card on the reader and click obtain to add RF card for the user. Path: **Intercom > User**.



### Parameter Set-up:

- **User ID:** enter the user ID. The user ID is 11 digits maximum in length and can not be reused for other users. The User ID can be generated automatically or manually.
- **Name:** enter the user name.
- **Code:** place the card on the device card reader area and click obtain.

**Note:**

- RF card with 13.56 MHz and 125 KHz can be applicable to the door phone for door access.

## 14.5. Edit the User-specific door access data

You can search user(s)-specific door access and edit the door access data on the web **Intercom > User** interface.

Index	Source	User ID	Name	Private PIN	RF Card	Floor No.	Web Relay	Schedule-Relay	Edit
<input checked="" type="checkbox"/> 1	Local	12334	Ryan	1221312	123123	0	0	1001-1;	
<input type="checkbox"/> 2									
<input type="checkbox"/> 3									

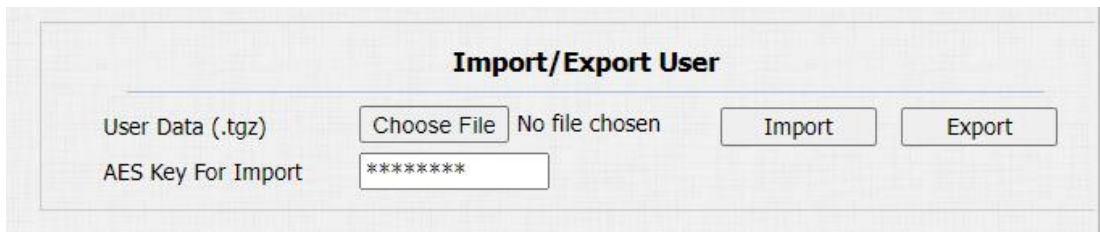
**Note:**

- Users synchronized from the SmartPlus can not be edited or deleted.

## 14.6. Import and Export User Data of Access Control

R28 series support User Data of access control to be shared among Akuvox

R28 series door phones through import and export while you can also export the facial data out of the door phone and then import it to a third-party device. To configure the configuration on the web **Intercom > User > Import/Export User** interface.



**Import/Export User**

User Data (.tgz)  No file chosen

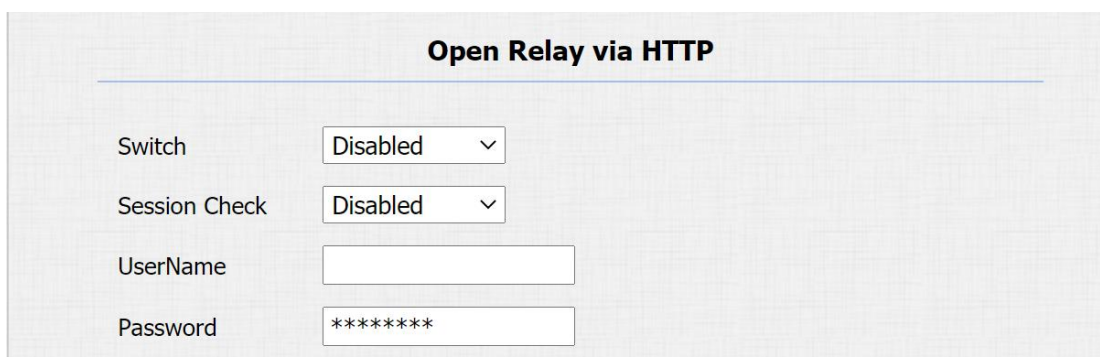
AES Key For Import

#### Parameter Set-up:

- **AES Key For Import:** enter the AES code before importing the AES-encrypted .tgz file to the door phone.

## 14.7. Configure Open Relay via HTTP for Door Unlock

You can unlock the door remotely without approaching the device physically for the door access by typing in the created HTTP command (URL) on the web browser to trigger the relay when you are not available by the door for the door access. To do this configuration on web **Intercom > Relay > Open Relay Via HTTP** interface.



**Open Relay via HTTP**

Switch  ▾

Session Check  ▾

UserName

Password

#### Parameter Set-up:

- **Switch:** enable the HTTP command to unlock the function by clicking on **Enable** field.

- **Session Check:** this feature is for some network security limitations, if you enable it, the door may not be unlocked in this way.
- **User Name:** enter the user name of the device web interface, for example, "Admin".
- **Password:** enter the password for the HTTP command. For example: "12345".

Please refer to the following example:

<http://192.168.35.127/fcgi/do?action=OpenDoor&UserName=admin&Password=12345&DoorNum=1>



**Note:**

- **DoorNum** in the HTTP command above refers to the relay number #1 to be triggered for the door access.

## 14.8. Configure Exit Button for Door Unlock

When you need to open the door from inside using the Exit button installed by the door, you can configure the door phone Input to trigger the relay for the door access on web **Intercom > Input** interface.

**Input**

---

**Input A**

---

Input Service

Trigger Option

Action To Execute FTP  Email  Sip Call  HTTP  Speed Dial

Http URL:

Action Delay  (0~300Sec)

Open Relay

Door Status DoorA: High

**Parameter Set-up:**

- **Input service:** select “**Enable** ” to be able to use the Input function.
- **Trigger Option:** select the trigger options according to the actual operation on the exit button.
- **Action To Execute:** select the method to carry out the action among six options: FTP, Email, HTTP, TFTP, Manager Dial and Speed Dial. Only R27X support Manager Dial and R28X supports Speed Dial.
- **Http URL:** enter the URL if you select the HTTP to carry out the action.
- **Action Delay:** set up the delay time when the action is carried out. For example, if you set the action delay time at 5 seconds., then the corresponding actions will be carried out 5 seconds after your press the button.
- **Open Relay:** set up relays to be triggered by the input.
- **Door Status:** display the status of the input signal.

## 14.9. Configure PIN Code for Door Unlock



You can create and modify both public PIN code and private PIN code for the door access on R28 door phone.

## 14.10. Configure Public Code for Door Unlock

Public PIN code is configured and used by the property in the same building or in the same community. To do this configuration on the web **Intercom > PIN Setting > Public PIN** interface.



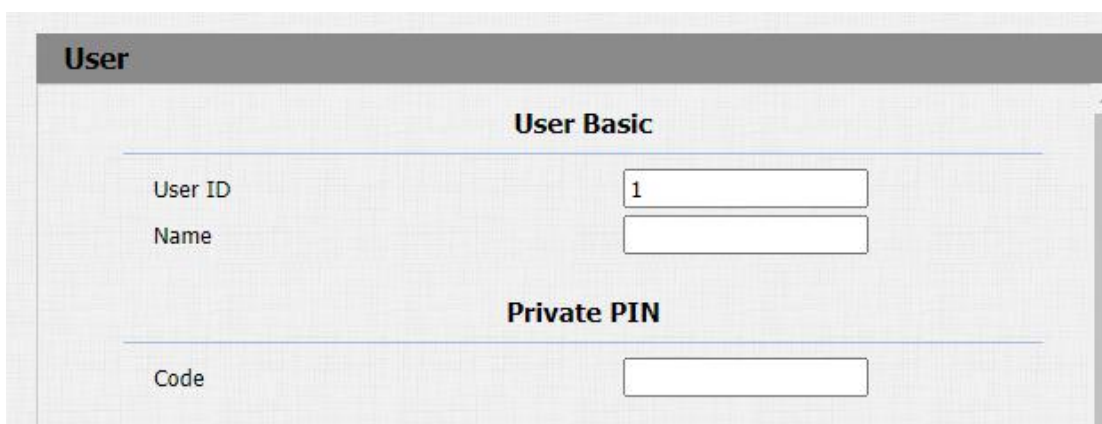
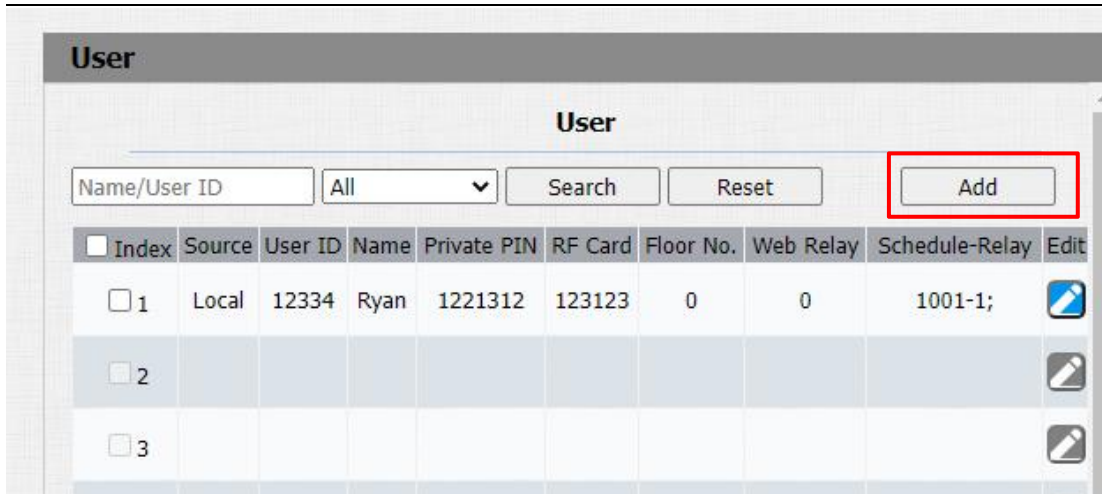
Public PIN	
Enabled	Disabled ▾
PIN Code	33333333 (3-8 digit number)

### Parameter Set-up:

- **Enabled:** Disable public key feature, the door can not be opened by public key. Enable it, you can use public key to open the door.
- **PIN Code:** customize 3-8 digit numbers for public key value.

### 14.10.1. Configure Private PIN Code on the Web Interface

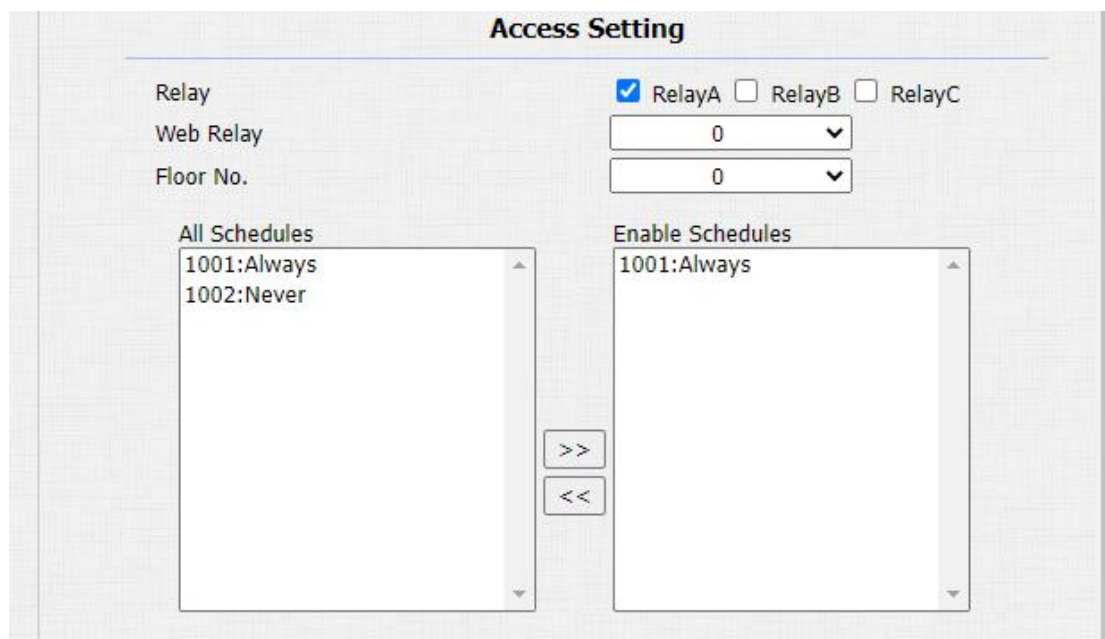
On the web interface, you can not only set up PIN code, but also set and select the door access schedule that you created for the validity of the PIN Code access during a certain time span you scheduled. Path: **Intercom > User**.



**Parameter Set-up:**

- **User ID:** enter user's ID.
- **Name:** enter the user name ( resident's name).
- **Code:** enter the user's private PIN.

After user information and PIN code are entered, you can scroll down to **Access Setting** and configure private PIN code access control.



**Parameter Set-up:**

- **Relay:** select the relay(s) that you want to apply the private PIN code for the door unlock.
- **Web relay:** select the specific number of web relay action commands you have set up on the web interface.
- **Floor NO:** enter the resident's floor number.
- **Schedule:** select from the created door access schedule on the right box and move the one to be applied to the user(s)-specific PIN code door access to the box on the right side.

You are required to enable the PIN code before you can get door access via private PIN code, you can navigate to **Intercom > PIN setting > Private PIN.**



## 15. Security

### 15.1. Tamper Alarm Setting

Tamper alarm function serves as a protection against any unauthorized removal of the devices by triggering off the temper alarm while sending out calls to the designated location. Tamper alarm will be triggered off when the door phone changes its gravity value as opposed to its original gravity value set up when the device is installed. To do this configuration on the web **Intercom > Advanced > Tamper Alarm** interface.

Tamper Alarm	
Tamper Alarm	Disabled ▾
Gravity Sensor Threshold	32 (0~127)

#### Parameter Set-up:

- **Tamper Alarm:** click to select "ON" in the Tamper Alarm field in order to enable the anti-theft alarm function.
- **Gravity Sensor Threshold:** set the threshold for the gravity sensory sensitivity. The lower the value is, the more sensitive the gravity sensor. The gravity sensor value is 32 by default.

### 15.2. Client Certificate Setting

Certificates can ensure communication integrity and privacy when deploying Akuvox door phone. So, when user needs to establish SSL protocol, it is necessary to upload corresponding certificates for verification.

**Web Server Certificate:** it is the certificate that sends to client for authentication when client require an SSL connection with Akuvox door phone. Currently, the format of certificate can be accepted by Akuvox door phone is \*.PEM file.

**Client Certificate:** When Akuvox door phone Phone required an SSL connection with server, the phone must verify the server to make sure it can be trusted. and the server will send its certificate to the Akuvox door phone. Then the door phone will verify this certificate according to client certificate list.

### 15.2.1. Configure Action of Input

When Input interface is working, it can also trigger an action. You can do this configuration on web **Intercom > Input** interface.

Action To Execute: FTP  Email  Sip Call  HTTP  Speed Dial

Http URL:

**Parameter Set-up:**

- **Action to execute:** To choose which action to execute after triggering.

### 15.2.2. Web Server Certificate

To upload Web Server certificate on the device web interface **Security > Advanced > Web Server Certificate**.

Index	Issue To	Issuer	Expire Time	Delete
1	IPphone	IPphone	Sun Oct 9 16:00:00 2034	Delete

**Web Server Certificate Upload(.PEM/.DER/.CER)**

Choose File No file chosen

### 15.2.3. Client Certificate

To upload and configure client certificate on the same page.

### Client Certificate

Index	Issue To	Issuer	Expire Time	
1				<input type="checkbox"/>
2				<input type="checkbox"/>
3				<input type="checkbox"/>
4				<input type="checkbox"/>
5				<input type="checkbox"/>
6				<input type="checkbox"/>
7				<input type="checkbox"/>
8				<input type="checkbox"/>
9				<input type="checkbox"/>
10				<input type="checkbox"/>

---

### Client Certificate Upload(.PEM/.DER/.CER/.CRT)

Index

Choose File

 No file chosen

Auto ▼

Submit

Cancel

Only Accept Trusted Certificates

Disabled ▼

**Parameter Set-up:**

- **Index:** select the desired value from drop-down list of Index. If you select **Auto** value, the uploaded certificate will be displayed by numeric order. If you select value from **1** to **10**, the uploaded certificate will be displayed according to the value that the user selected.
- **Select File:** click Choose file browse local drive, and locate the desired certificate. (\*.pem only)
- **Only Accept Trusted certificates:** if you select **Enabled**, as long as the authentication success, the phone will verify the server certificate based on the client certificate list. If you select **Disabled**, the phone will not verify the server certificate no matter whether the certificate is valid or not.

## 15.3. Motion Detection

Motion Detection is often used for unattended surveillance video and automatic alarm. The images collected by the camera at different frame rates will be calculated and compared by the CPU according to a certain algorithm. When the picture changes, if someone walks by, the lens is moved, the number obtained by the calculation and comparison result will exceed the threshold and indicate that the system can the corresponding processing is made automatically.

### 15.3.1. Configure Motion Detection

You can turn on the motion detection and set up the motion detection interval on the web **Intercom > Motion** interface.

**Motion Detection Options**

---

Motion Detection Disabled ▾

Time 10 (0~120 Sec)

**Motion Detect Time Setting**

---

Mon  Tue  Wed  Thur

Fri  Sat  Sun  Check All

00 ▾ : 00 ▾ - 23 ▾ : 59 ▾

#### Parameter Set-up:

- **Motion Detection:** To enable or disable Motion Detection.
- **Time:** set the time interval for the motion detection. If you set the default time interval as "10" Sec, then the motion detection time span will be 10 seconds. Assuming that we set the time interval as "10" then, and the first movement captured can be seen as the start point of the motion detection, and if the movement continues through 7 seconds of the 10 second interval, then the alarm will be triggered at 7 seconds ( the first

trigger point) and motion detection action can be triggered (sending out notification) any where between **7-10** seconds once the movement is detected. "10" Sec interval is a complete cycle of the motion detection before it starts another cycle of the same time interval. To be more specific, the first trigger point can be calculated as the "**Time interval minus three**".

## 15.4. Security Notification Setting

### 15.4.1. Email Notification Setting

If you want to receive the security notification via email, you can configure the Email notification on the web **Intercom > Action > Email Notification** interface properly. The email notification will show as the captures.

Email Notification	
Sender's email address	<input type="text" value="neil.fang1214@gmail.com"/>
Receiver's email address	<input type="text" value="neil.fang@akuvox.com"/>
SMTP server address	<input type="text" value="smtps://smtp.gmail.com"/>
SMTP user name	<input type="text" value="neil.fang1214@gmail.com"/>
SMTP password	<input type="password" value="••••••"/>
Email subject	<input type="text" value="Test"/>
Email content	<input type="text" value="Only for Testing."/>
<input type="button" value="Email Test"/>	

#### Parameter Set-up:

- **Sender's email address:** enter the sender's email address from which the email notification will be sent out.
- **Receiver's Email Address:** enter the receiver's email address.



- **SMTP Server Address:** enter the SMTP server address of the sender.
- **SMTP User Name:** enter the SMTP user name, which is usually the same with sender's email address.
- **SMTP Password:** configure the password of SMTP service, which is same with sender's email address.
- **Email Subject:** enter the subject of the email.
- **Email Content:** compile the contents of emails according to your need.

## 15.4.2. FTP Notification Setting

If you want to receive the security notification via FTP, you can configure the FTP notification on the web **Intercom > Action > FTP Notification** properly.

FTP Notification	
FTP Server	<input type="text" value="192.168.1.155"/>
FTP User Name	<input type="text" value="admin"/>
FTP Password	<input type="password" value="....."/>
	<input type="button" value="FTP Test"/>

### Parameter set-up:

- **FTP Server:** enter the address (URL) of the FTP server for the FTP notification.
- **FTP User Name:** enter the FTP server user name.
- **FTP Password:** enter the FTP server password.

## 15.4.3. SIP Call Notification Setting

In addition to FTP and Email notification, the door phone can also make a SIP call when some feature action is triggered. To configure a SIP call notification on web **Intercom > Action > SIP Call Notification** interface.

**SIP Call Notification**

---

SIP Call Number

SIP Caller Name

**Parameter Set-up:**

- **SIP Call Number:** to configure the SIP call number.
- **SIP Call Name:** to configure the display name of door phone.

### 15.4.4. Call Event Notification

Enable this feature if you want to be notified when any outgoing calls from the door phone are not answered. The notification is made via FTP, Email, and HTTP. On the web, navigate to **Intercom > Basic > Call Event**.

**Call Event**

---

No Answer Action

Action To Execute      FTP     Email     Http

Http URL:

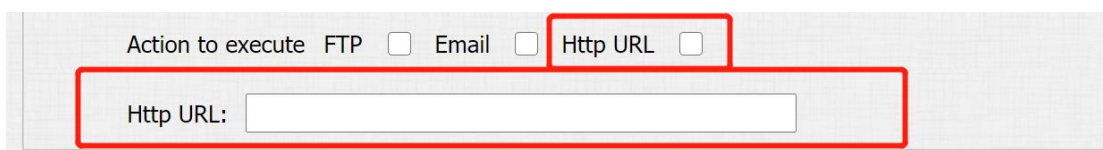
**Parameter Set-up:**

- **No Answer Action:** enable the feature if needed.
- **Action To Execute:** select FTP, Email or HTTP method for the notification. If FTP is selected, a screenshot of the caller will be sent via as notification. If Email is selected, the Emails containing a screenshot of the caller will be sent. If HTTP is selected, you can add the event message to the HTTP URL before sending.

- **Http URL:** enter the HTTP URL that will be sent to the Http server. For example:http//192.168.31.6/door phone#1.

### 15.4.5. HTTP URL Notification Configuration

Akuvox door phones support sending the HTTP notification to the third party when some features are triggered. HTTP notification can be set up in specific chapters, please check [chapter 16.4](#). The URL format: **http://http server IP address/any information.**



The screenshot shows a configuration panel with three radio buttons: 'FTP', 'Email', and 'Http URL'. The 'Http URL' radio button is selected. Below these buttons is a text input field labeled 'Http URL:'. A red rectangular box highlights the 'Http URL' radio button and the input field.

#### Parameter Set-up:

- **Http URL:** tick the check box to enable HTTP URL notification.
- **HTTP URL:** If you choose HTTP mode, enter the URL format: **http://http server IP address/any information.**

## 15.5.Action URL

R28 allows you to set up specific HTTP URL command that will be sent to the HTTP server for the predefined actions. Relevant actions will be initiated if there occurs any changes in the relay status, input status, PIN code, and RF card access for security purpose. Path: **Phone > Actions URL.**

**Action URL**

**Action URL**

Active	<input style="width: 95%;" type="text" value="Disabled"/>
Make Call	<input style="width: 95%;" type="text"/>
Hang Up	<input style="width: 95%;" type="text"/>
RelayA Triggered	<input style="width: 95%;" type="text"/>
RelayB Triggered	<input style="width: 95%;" type="text"/>
RelayC Triggered	<input style="width: 95%;" type="text"/>
RelayA Closed	<input style="width: 95%;" type="text"/>
RelayB Closed	<input style="width: 95%;" type="text"/>
RelayC Closed	<input style="width: 95%;" type="text"/>
InputA Triggered	<input style="width: 95%;" type="text"/>
InputB Triggered	<input style="width: 95%;" type="text"/>
InputC Triggered	<input style="width: 95%;" type="text"/>
InputA Closed	<input style="width: 95%;" type="text"/>
InputB Closed	<input style="width: 95%;" type="text"/>
InputC Closed	<input style="width: 95%;" type="text"/>
Valid Code Entered	<input style="width: 95%;" type="text"/>
Invalid Code Entered	<input style="width: 95%;" type="text"/>
Valid Card Entered	<input style="width: 95%;" type="text"/>
Invalid Card Entered	<input style="width: 95%;" type="text"/>

## 15.6. Security Action Configuration

### 15.6.1. Configure Action of Call

When pressing the push button, the door phone will trigger the pre-configured action type, the notification can be sent out by Email, FTP notification or SIP call. To do this configuration on web **Intercom > Basic** interface.

**Call Event**

---

No Answer Action Disabled ▾

Action To Execute FTP  Email  Http

Http URL:

**Parameter Set-up:**

- **No Answer Action:** if the call will not be answered, it still trigger the action event after enabling this feature.
- **Action to execute:** to choose which action to be executed after triggering.

### 15.6.2. Configure Action of Motion

When the Motion Detection feature is working , you can make it trigger an action. To do this configuration on web **intercom > Motion** interface.

**Action to execute**

---

Action to execute FTP  Email  Sip Call  HTTP

Http URL:

**Parameter Set-up:**

- **Action to execute:** To choose which action to be executed after triggering.

### 15.6.3. Configure Action of Body Temperature

When the detected wrist temperature is higher than the normal body temperature, it will trigger the SIP/IP call if you tick the checkbox of **Action To Execute**. To do this configuration on the web **Intercom > Body**

Temperature interface.

### Measuring Body Temperature

---

Mode Disabled ▾

Temperature Unit Fahrenheit ▾

Normal Body Temperature 99.14 (Below99.14°F)

(If the detected temperature is lower than 95.00°F , the device will prompt low temperature, please try again later)

Action To Execute  SIP/IP Call

SIP/IP Call Number 111

**Parameter Set-up:**

- **Action to execute:** to choose which action to execute after triggering.
- **SIP/IP Number:** enter the SIP/IP Number you need. The SIP number here is different from the SIP number in Action interface. This SIP number is only available when body temperature is triggered.

## 15.7.Voice Encryption

SRTP(Secure Real-time Transport Protocol) is a protocol defined on the basis of Real-time Transport Protocol. The data of the transmission protocol provides encryption, message authentication, integrity assurance and replay protection. To configure this feature on web **Account > Advanced > Encryption** interface.

### Encryption

---

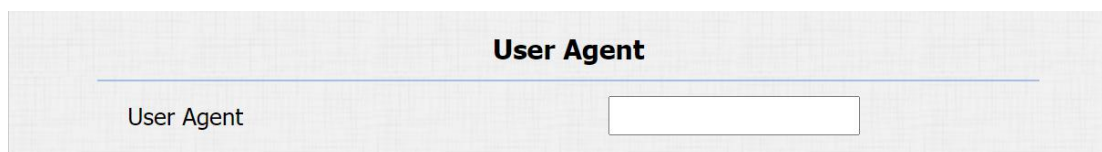
Voice Encryption(SRTP) Disabled ▾

**Parameter Set-up:**

- **Voice Encryption(SRTP):** choose **Disabled**, **Optional** or **Compulsory** for SRTP. If it is **Optional** or **Compulsory**, the voice during the call is encrypted, and you can grab the RTP packet to view.

## 15.8.7 User Agent

You can customize user agent field in the SIP message. If user agent is set to specific value, users can see the information from PCAP. If user agent is blank, by default, users can see the company name "Akuvox", model number and firmware version from PCAP. To do this configuration on the web **Account > Advanced > User Agent** interface.



### Parameter Set-up:

- **User Agent:** support to enter another specific value, Akuvox is by default.

## 15.9. Body Temperature

R28A provides you with an optional wrist temperature measurement function designed to be applied in the situation where the measurement becomes necessary for the safety of the residents and visitors etc. Residents and visitors are required to go through temperature measurement before they are allowed the door access. Before entering the configuration, you need to correctly connect the MD01 external temperature measurement module to cooperate with the software's wrist temperature function. The MD01 module can be connected normally through the RS485A, RS485B, 12V Power Output, and GND interfaces of the docking device.

**Note:** This feature need to be worked with extra wrist temperature detection sensor Akuvox MD01, please consult Akuvox technical team for this model in detail.



Akuvox MD01

### Measuring Body Temperature

---

Mode	<input type="text" value="Disabled"/>	▼
Temperature Unit	<input type="text" value="Fahrenheit"/>	▼
Normal Body Temperature	<input type="text" value="99.14"/>	(Below 99.14°F)

(If the detected temperature is lower than 95.00°F , the device will prompt low temperature, please try again later)

#### Parameter Set-up:

- **Mode:** there are two drop-down options, "Disabled" and "Wrist", select "Disabled", the device displays the home page normally (display call, contact, pin, security); select "Wrist", the device will open the wrist temperature test function.
  
- **Temperature Unit:** there are two drop-down options, "Fahrenheit" and "Centigrade". The value displayed by the device will change with the selection, and the value in the corresponding prompt will also change. The conversion formula between Fahrenheit and Celsius:  $^{\circ}\text{C} = 5/9 \times (^{\circ}\text{F} - 32)$ .
  
- **Normal Body Temperature:** the default setting is 99.14 degrees Fahrenheit (37.3 degrees Celsius). If the detected temperature is lower than 95.00 degrees Fahrenheit (35 degrees Celsius), the device will prompt low temperature, please try again later.



## 16. Monitor and Image

### 16.1. RTSP Stream Monitoring

Akuvox door phones support RTSP stream that allows intercom devices such as indoor monitor or the monitoring unit from the third party to monitor or obtain the real time audio/ video (RTSP stream) from the door phone using the correct URL.

#### 16.1.1. RTSP Basic Setting

You are required to set up RTSP function on device web **Intercom > RTSP > RTSP Basic** interface in terms of RTSP Authorization, authentication and password, etc before you are able to use the function.

**RTSP Basic**

RTSP Server Enabled	<input checked="" type="checkbox"/>
RTSP Authorization	<input type="checkbox"/>
MJPEG Authorization	<input type="checkbox"/>
RTSP Authentication Type	Basic ▾
RTSP User Name	admin
RTSP Password	••••••••

#### Parameter Set-up:

- **RTSP Server Enable:** click on Enable and Disable in **RTSP Enable** field to turn on or turn off the RTSP function.
- **RTSP Authorization:** click on Enable and Disable in RTSP Authorization field to enable or disable the RTSP authorization. If you enable the RTSP Authorization, you are required to enter RTSP Authentication Type, RTSP Username, RTSP Password on the intercom device such as indoor monitor for authorization.

- **RTSP User Name:** enter the name used for RTSP authorization.
- **RTSP User Password:** enter the password for RTSP authorization.
- **RTSP Authentication Type:** select RTSP authentication type between "Basic" and "Digest". "Basic" is the default authentication type.

### 16.1.2. RTSP OSD Setting

This feature is used to add watermark to the RTSP video or picture. To protect the owner of the video or image. To do this configuration on the web **Intercom > RTSP > RTSP OSD Setting** interface.

**RTSP OSD Setting**

---

RTSP OSD Color	<input style="width: 100%;" type="text" value="White"/> ▾
RTSP OSD Text	<input style="width: 100%;" type="text"/>

**Parameter Set-up:**

- **RTSP OSD Color:** there are five color options - White, Black, Red, Green, Blue for RTSP watermark text.
- **RTSP OSD Text:** enter the customized text you want to show for the watermark.

### 16.1.3. RTSP Stream Setting

You can select the video codec format for the RTSP stream for the monitoring and configure video resolution and bit-rate etc based on your actual network environment on the web **Intercom > RTSP > RTSP stream** interface.

RTSP Stream	
RTSP Audio Enabled	<input checked="" type="checkbox"/>
RTSP Video Enabled	<input checked="" type="checkbox"/>
RTSP Video2 Enabled	<input checked="" type="checkbox"/>
RTSP Audio Codec	PCMU ▾
RTSP Video Codec	H.264 ▾
RTSP Video2 Codec	H.264 ▾

**Parameter Set-up:**

- **RTSP Audio Enabled:** tick to enable RTSP audio which means, the door phone can also send audio information to the monitor by RTSP.
- **RTSP Video Enabled:** the door phone can send the video information to the monitor. After enabling RTSP feature, the video RTSP is enabled by default and can not be modified.
- **RTSP Video2 Enabled:** Akuvox door phones support 2 RTSP streams, you can enable the second one.
- **RTSP Audio Codec:** choose a suitable audio codec for RTSP audio.
- **RTSP Video/Video2 Codec:** choose a suitable video codec for RTSP video.

H.264 And H.265 Video Parameters	
Video Resolution	720P ▾
Video Framerate	30 fps ▾
Video Bitrate	2048 kbps ▾
Video2 Resolution	VGA ▾
Video2 Framerate	30 fps ▾
Video2 Bitrate	512 kbps ▾

**Parameter Set-up:**

- **Video Resolution:** select video resolutions among seven options: "QCIF",

"QVGA", "CIF", "VGA", "4CIF", "720P". The default video resolution is "4CIF". and the video from the door phone might not be able to be shown in the indoor monitor if the resolution is set higher than "4CIF".

- **Video Framerate:** "30fps" is the video frame rate by default.
- **Video Bitrate:** select video bit-rate among six options: "128 kbps", "256kbps", "512 kbps", "1024 kbps", "2048 kbps", "4096 kbps" according to your network environment. The default video bit-rate is "2048 kbps".
- **Video2 Resolution:** select video resolution for the second video stream channel. While the default video solution is "VGA".
- **Video2 Framerate:** select the video framerate for the second video stream channel. "25fps" is the video frame rate by default for the second video stream channel.
- **Video2 Bitrate:** select video bit-rate among the six options for the second video stream channel. While the second video stream channel is "512 kbps" by default.

## 16.2.MJPEG Image Capturing

Akuvox door phones allow you to capture the Mjpeg format monitoring image if needed. You can enable the Mjpeg function on **Intercom > RTSP > RTSP Basic** and set the image quality on the web **Intercom > RTSP > MJPEG Video Parameters** interface.

The screenshot shows the 'RTSP Basic' configuration page. It contains several settings:

Setting	Value
RTSP Server Enabled	<input checked="" type="checkbox"/>
RTSP Authorization	<input type="checkbox"/>
MJPEG Authorization	<input type="checkbox"/>
RTSP Authentication Type	Basic
RTSP User Name	admin
RTSP Password	.....

The 'MJPEG Authorization' row is highlighted with a red rectangular border.

### MJPEG Video Parameters

---

Video Resolution	CIF <span style="float: right;">▼</span>
Video Framerate	25 fps <span style="float: right;">▼</span>
Video Quality	90 <span style="float: right;">▼</span>

**Parameter Set-up:**

- **MJPEG Authorization:** tick it to access device video or real-time screenshots through a browser (http address such as: http://device IP:8080/video.cgi (dynamic video), http://device IP:8080/jpeg.cgi (static screenshot) )
- **Video Resolution:** select video resolutions among seven options: "QCIF", "QVGA", "CIF", "VGA", "4CIF", "720P". The default video resolution is "4CIF". And the video from the door phone might not be able to be shown on the indoor monitor if the resolution is set higher than "4CIF".
- **Video Framerate:** 30fps is the video frame rate by default.
- **Video Quality:** the video bitrate, from 50 to 90.

## 16.2.1. NACK

NACK (**Negative Acknowledgement**) used to ensure the smooth and continued data transmission for the video call. To enable NACK, navigate to **Phone > Call Feature > Others**.

### Others

---

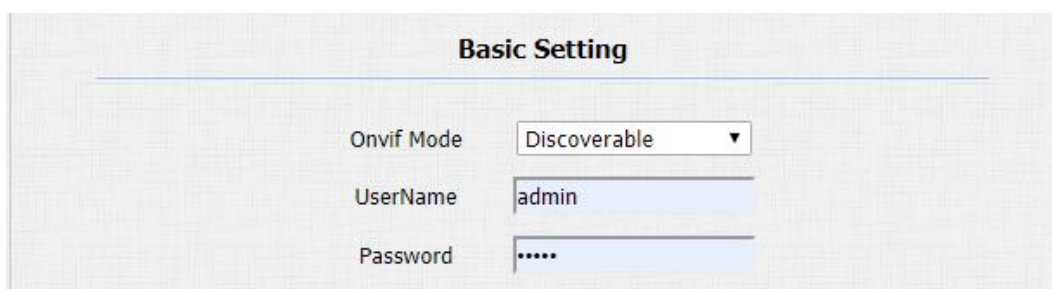
Return Code When Refuse	486(Busy Here) <span style="float: right;">▼</span>
Auto Answer Delay	0 <span style="float: right;">(0~5Sec)</span>
Auto Answer Mode	Video <span style="float: right;">▼</span>
Direct IP	Enabled <span style="float: right;">▼</span>
Direct IP AutoAnswer	Enabled <span style="float: right;">▼</span>
Direct IP Port	5060 <span style="float: right;">(1~65535)</span>
NACK	Disabled <span style="float: right;">▼</span>

**Parameter Set-up:**

- **NACK Enabled:** enable the NACK. It can be used to prevent losing data packet in the weak network environment when discontinued and mosaic video image occurred.

## 16.3.ONVIF

Real-time video from the door phone camera can be searched and obtained by the Akuvox indoor monitor or by third party devices such as NVR (**Network Video Recorder**) you can configure the ONVIF function on the web **Intercom > ONVIF** interface so that other devices will be able to see the video from the door phone.



The screenshot shows a web interface titled "Basic Setting" for ONVIF configuration. It contains three fields: "Onvif Mode" with a dropdown menu set to "Discoverable", "UserName" with a text input field containing "admin", and "Password" with a masked text input field showing ".....".

**Parameter Set-up:**

- **Discoverable:** tick the check box to enable the Discoverable ONVIF mode. If you select "**Discoverable**" then the video from the door phone camera can be searched by other devices.
- **User Name:** enter the user name. The user name is "**admin**" by default.
- **Password:** enter the password. The password is "**admin**" by default.

After the setting is complete, you can enter the ONVIF URL on the third party device to view the video stream.

For example: **http://IP address:80/onvif/device\_service**

**Note:**

- Fill in the specific IP address of the door phone in the URL.

## 16.4.Live Stream

If you want to check the real-time video from the door phone, you can go to the device web **Intercom > Live Stream** interface to obtain the real-time video or you can also enter the correct URL on the web browser to obtain it directly. To check the real time video using URL, you can Enter the correct URL ([http://IP\\_address:8080/video.cgi](http://IP_address:8080/video.cgi)) on the web browser if you want to obtain the real-time video directly instead of going to the web interface.



# 17. Logs

## 17.1.Call Logs

If you want to check on the calls inclusive of the dial-out calls, received calls and missed calls in a certain period of time, you can check and search the call log on the device web **Phone > Call Log** interface.

Call History						
			All	▼	Hang Up	
Index	Type	Date	Time	Local Identity	Name	Number
1	Dialed	2021-02-04	09:30:33	192.168.31.2 @192.168.31.2	Unknown	<a href="#">192.168.35.1</a> <a href="#">12@192.168.35.112</a>
2	Received	2021-02-04	09:29:57	192.168.31.2 @192.168.31.2	192.168.35.112	<a href="#">192.168.35.1</a> <a href="#">12@192.168.35.112</a>
3	Dialed	2021-02-04	09:29:06	192.168.31.2 @192.168.31.2	Unknown	<a href="#">192.168.35.1</a> <a href="#">12@192.168.35.112</a>

**Parameter Set-up:**

- **Call History:** select call history among four options: **“All”, “Dialed” “Received” “Missed”** for the specific type of call log to be displayed.
- **Hangup:** to hangup the call from web.
- **Index:** the order of the call logs.
- **Date:** the date for the call log.
- **Time:**the time for the call log.
- **Name/Number:** the name and number of the contact.



## 17.2. Door Logs

If you want to search and check and import/export the various types of door access history, you can search and check the door logs on the device web **Phone > Door Log** interface.

### Door Log

Index	Name	Code	Type	Date	Time	Status	
1	Unknown	01320C39	Card	2021-05-13	03:50:55	Failed	<input type="checkbox"/>
2							<input type="checkbox"/>
3							<input type="checkbox"/>
4							<input type="checkbox"/>
5							<input type="checkbox"/>
6							<input type="checkbox"/>
7							<input type="checkbox"/>
8							<input type="checkbox"/>
9							<input type="checkbox"/>
10							<input type="checkbox"/>
11							<input type="checkbox"/>
12							<input type="checkbox"/>
13							<input type="checkbox"/>
14							<input type="checkbox"/>
15							<input type="checkbox"/>

Page: 1

### Import/Export Door Log(.xml)

No file chosen

### Parameter Set-up:

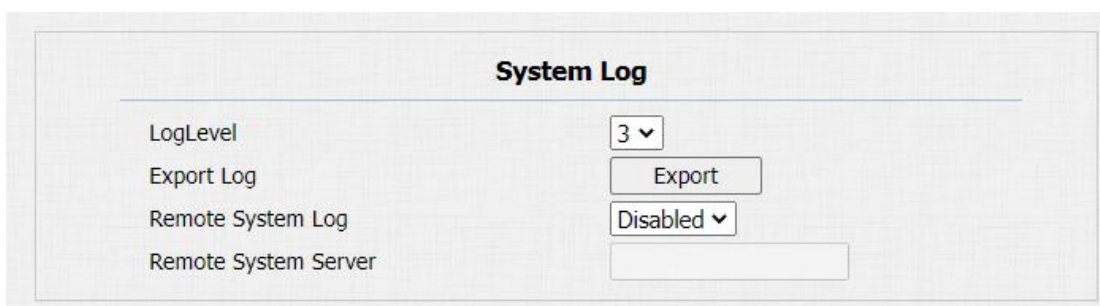
- **Index:** the order of the call logs.
- **Name:** if it is a locally added key or card, the corresponding added name will be displayed. If it is an unknown key or card, it will display Unknown.
- **Code:** if opening the door via PIN code, the corresponding PIN code will be displayed. If opening the door via RF cards, the corresponding card number will be displayed, and if the door is opened by HTTP command, it will be empty.

- **Type:** if opening the door via PIN code, **Password** will be displayed. If opening the door via RF cards, **Card** will be displayed, and if the door is opened by HTTP command, **Http** will be displayed.
- **Date:** the date for opening the door.
- **Time:** the time for opening the door.
- **Status:** the door opening result **Success** or **Failed**.

## 18. Debug

### 18.1. System Log

System log in the door phone can be used for debugging purpose. If you want to export the system out to a local PC or to a remote server for debugging , you can set up the function on the web **Upgrade > Advanced > System Log** interface.



System Log	
LogLevel	3 ▾
Export Log	Export
Remote System Log	Disabled ▾
Remote System Server	

#### Parameter Set-up:

- **LogLevel:** select log levels from 1 to 7 levels. You will be instructed by Akuvox technical staff about the specific log level to be entered for debugging purpose. The default log level is "3". the higher the level is, the more complete the log is.
- **Export Log:** click the **Export** tab to export temporary debug log file to a local PC.
- **Remote System Log:** select "Enable" or "Disable" if you want to enable or disable the remote system log.
- **Remote System Server:** enter the remote server address to receive the device log. And the remote server address will be provided by Akuvox technical support.

## 18.2.Remote Debug Server

Remote debug server is used to access the log remotely for debugging purpose. On the web,\. Navigate to **Intercom > Advanced > Remote Debug Server**.

### Parameter Set-up:

- **Service:** disable or enable the remote debug server.
- **Connect Status:** displays the remote debug server connection status.
- **IP:** enter the remote debug server IP address.
- **Port:** enter the remote debug server port.

## 18.3.PCAP

PCAP in Akuvox door phone is used to capture the data package going in and out of the devices for debugging and troubleshooting purpose. You can set up the PCAP on the device web **Upgrade > Advanced > PCAP** interface properly before using it.

**Parameter Set-up:**

- **Specific Port:** select the specific ports from 1-65535 so that only the data packet from the specific port can be captured. You can leave the field blank by default.
- **PCAP:** click **Start** tab and **Stop** tab to capture a certain range of data packets before clicking **Export** tab to export the data packets to your Local PC.
- **PCAP Auto Refresh:** select "**Enable**" or "**Disable**" to turn on or turn off the PCAP auto refresh function. If you set it as "Enable" then the PCAP will continue to capture data packet even after the data packets reached their 1M maximum in capacity. If you set it as "**Disable**" the PCAP will stop data packet capturing when the data packet captured reached the maximum capturing capacity of 1MB.

## 19. Firmware Upgrade

Firmwares of different versions for Akuvox door phone can be upgraded on the device web **Upgrade > Basic** interface.

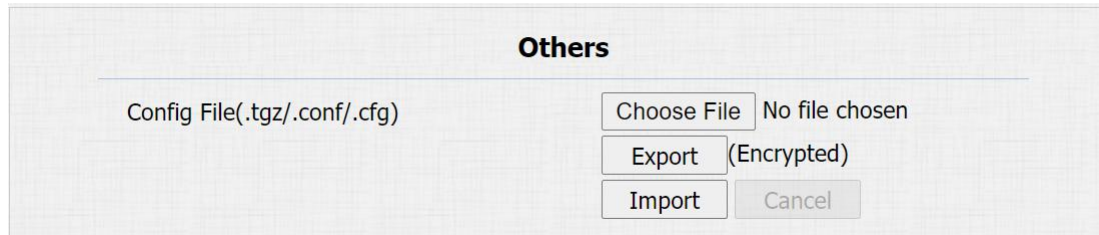
Firmware Version	28.30.2.215
Hardware Version	28.0
Upgrade	<input type="button" value="Choose File"/> No file chosen
	Reset: <input type="checkbox"/>
	<input type="button" value="Submit"/> <input type="button" value="Cancel"/>

### Parameter Set-up:

- **Upgrade:** Choose.rom firmware from your PC, then click **Submit** to update.

## 20. Backup

Configuration files can be imported to or exported out of the device to your local PC on the device web **Upgrade > Advanced > Others** interface if needed.



The screenshot shows a web interface titled "Others". Below the title is a horizontal line. Underneath, there is a text label "Config File(.tgz/.conf/.cfg)". To the right of this label are four buttons: "Choose File" (with "No file chosen" text to its right), "Export (Encrypted)", "Import", and "Cancel".

### Parameter Set-up:

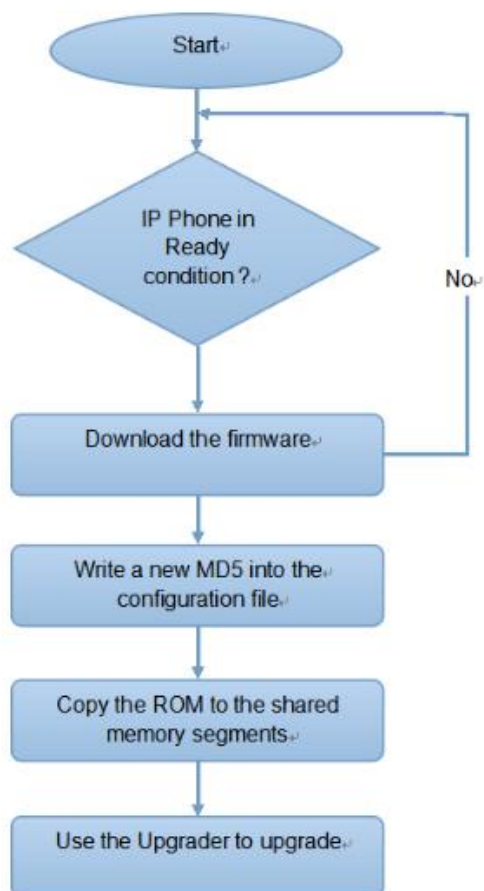
- **Export Config File:** to export the current config file.
- **Export/Import:** to export the current config file (Encrypted) or import new config file.

## 21. Auto-provisioning

Configurations and upgrading on Akuvox door phone can be done on the web interface via one-time auto-provisioning and scheduled auto-provisioning via configuration files, thus saving you from setting up configurations needed one by one manually on the door phone.

### 21.1. Provisioning Principle

Auto-provisioning is a feature used to configure or upgrade the devices in batch via third party servers. **DHCP, PNP, TFTP, FTP, HTTPS** are the protocols used by the Akuvox intercom devices to access the URL of the address of the third party server which stores configuration files and firmware, which will then be used to update the firmware and the corresponding parameters on the door phone.





## 21.2. Configuration Files for Auto-provisioning

Configuration files have two formats for auto-provisioning. One is general configuration files used for the general provisioning and another one is the MAC-based configuration provisioning.

The difference between the two types of configuration files is shown below:

- **General configuration provisioning:** a general file is stored in a server from which all the related devices will be able to download the same configuration file to update parameters on the devices. For example : r000000000020.cfg.
- **MAC-based configuration provisioning:** MAC-based configuration files are used for the auto-provisioning of a specific device as distinguished by its unique MAC number. And the configuration files named with device MAC number will be matched automatically with the device MAC number before being downloaded for the provisioning on the specific device.

To get the Autop configuration file template on **Upgrade > Advanced > Automatic Autop** interface.

**Automatic Autop**

Mode	Power On	▼
Schedule	Sunday	▼
	22	Hour(0~23)
	0	Min(0~59)
Clear MD5	Submit	
Export Autop Template	Export	



**Note:**

- If a server has these two types of configuration files, then IP devices will first access the general configuration files before accessing the MAC-based configuration files.

## 21.3.AutoP Schedule

Akuvox provides you with different Autop methods that enable the door phone to perform provisioning for itself at a specific time according to your schedule.

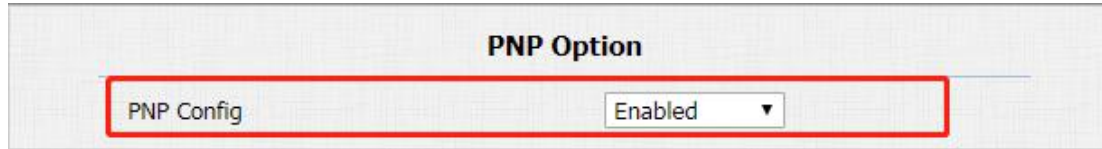
Automatic Autop	
Mode	Power On <input type="button" value="v"/>
Schedule	Sunday <input type="button" value="v"/>
	22 <input type="button" value="v"/> Hour(0~23)
	0 <input type="button" value="v"/> Min(0~59)

**Parameter Set-up:**

- **Mode:** select "**Power on**", "**Repeatedly**", "**Power On + Repeatedly**", "**Hourly Repeat**" as your Autop schedule.  
 Select "**Power on**", if you want the device to perform Autop every time it boots up.  
 Select "**Repeatedly**", if you want the device to perform Autop according to the schedule you set up.  
 Select "**Power On + Repeatedly**", if you want to combine **Power On** Mode and **Repeatedly** mode that will enable the device to perform Autop every time it boots up or according to the schedule you set up.  
 Select "**Hourly Repeat**", if you want the device to perform Autop every hour.

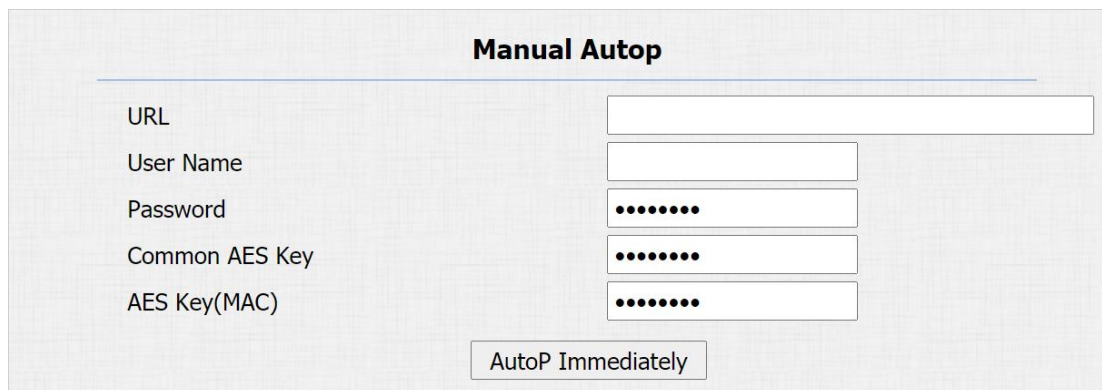
## 21.4.PNP Configuration

Plug and Play (PNP) is a combination of hardware and software support that enables a computer system to recognize and adapt to hardware configuration changes with little or no intervention by a user. To do this configuration on web **Upgrade > Advanced > PNP Option** interface.



## 21.5.Static Provisioning Configuration

You can manually set up a specific server URL for downloading the firmware or configuration file. If an autop schedule is set up, the door phone will perform the auto provisioning on a specific timing according to Autop schedule you set up. In addition, TFTP, FTP, HTTP, and HTTPS are the protocols that can be used for upgrading the device firmware and configuration.



### Parameter set-up:

- **URL:** set up tftp, http, https, ftp server address for the provisioning
- **User Name:** set up a user name if the server needs a user name be accessed to otherwise leave it blank.
- **Password:** set up a password if the server needs a password be accessed to otherwise leave it blank.
- **Common AES Key:** set up AES code for the intercom to decipher the

general Auto Provisioning configuration files.

- **AES Key (MAC):** set up AES code for the intercom to decipher the MAC-based auto provisioning configuration file.

 **Note:**

- AES is one type of encryption, it should be configured only when the config file is encrypted with AES, otherwise leave the field blank.

 **Note:**

**Server Address format:**

- TFTP: tftp://192.168.0.19/
- FTP: ftp://192.168.0.19/ (allows anonymous login)
- ftp://username:password@192.168.0.19/(requires a user name and password)
- HTTP: http://192.168.0.19/ (use the default port 80)
- http://192.168.0.19:8080/ (use other ports, such as 8080)
- HTTPS: https://192.168.0.19/ (use the default port 443)

 **Note:**

- Akuvox do not provide user specified server.
- Please prepare TFTP/FTP/HTTP/HTTPS server by yourself.

## 22. Integration with Third Party Device

### 22.1. Integration via Wiegand

If you want to integrate Akuvox door phone with the third party devices via Wiegand, you can configure the Wiegand on the web **Intercom > Wiegand** interface.

The screenshot shows the 'Wiegand Setting' interface. It features a title bar 'Wiegand Setting' and a sub-header 'Wiegand'. Below this, there are seven configuration items, each with a label and a dropdown menu:

- Wiegand Display Mode: 8HN
- Wiegand Card Reader Mode: Wiegand-26
- Wiegand Transfer Mode: Input
- Wiegand Input Data Order: Normal
- Wiegand Output Basic Data Order: Normal
- Wiegand Output Data Order: Normal
- Wiegand Output CRC: ON

#### Parameter Set-up:

- **Wiegand Display Mode:** select Wiegand Card code format among 8H10D; 6H3D5D; 6H8D; 8HN; 8HR; RAW.
- **Wiegand Card Reader Mode:** set the wiegand data transmission format among three options: "Wiegand 26", "Wiegand 34", "Wiegand 58". The transmission format should be identical between the door phone and the device to be integrated.
- **Wiegand Transfer Mode:** select Input, Output, Convert to Card No. Output Wiegand. if the door phone is used as a receiver, then set it as "Input" for the door phone. Select "Output" If you want wiegand output to be converted to card number before sending it from the door phone to a receiver. For facial recognition access, the user card number corresponding to the facial recognition access will be sent out in a binary system.

- **Wiegand Input Data Order:** set the Wiegand input data sequence between "Normal" and "Reversed" if you select "Reversed" then the input card number will be reversed and vice versa.
- **Wiegand Output Data Order:** set the Wiegand output data sequence between "Normal" and "Reversed" if you select "Reversed" then the input card number will be reversed and vice versa.
- **Wiegand Output CRC:** tick to enable the parity check function to ensure that signal-based data can be transmitted correctly according to the established data transmission format.
- **RF Card Verification:** enable the RF card verification if needed.

You can configure the wiegand output mode if needed. The output occurs when you press the PIN code on the device.



The screenshot shows a dialog box titled "Forward To Wiegand Output". Inside the dialog, there is a label "PIN" followed by a dropdown menu that is currently set to "Disabled". Below the dropdown menu, there are two buttons: "Submit" and "Cancel".

## 22.2. Integration via HTTP API

HTTP API is designed to achieve a network-based integration between the third party device with the Akuvox intercom device. You can configure the HTTP API function on the web **Intercom > HTTP API** interface for the integration.

**HTTP API**

**HTTP API**

---

HTTP API	Enabled <input type="button" value="v"/>
Auth Mode	Digest <input type="button" value="v"/>
User Name	admin
Password	*****
IP01	<input type="text"/>
IP02	<input type="text"/>
IP03	<input type="text"/>
IP04	<input type="text"/>
IP05	<input type="text"/>

**Parameter Set-up:**

- **HTTP API:** select **“Enable”** or **“Disable”** to enable or disable the HTTP API function for the third party integration. For example, if the function is disabled any request to initiate the integration will be denied and be returned HTTP 403 forbidden status.
- **Auth Mode:** select among four options: **“None”** **“WhiteList”** **“Basic”**, **“Digest”** for authorization type, which will be explained in detail in the following chart.
- **User Name:** enter the user name when **“Basic”** and **“Digest”** authorization mode is selected. The default user name is **“Admin”**.
- **Password:** enter the password when **“Basic”** and **“Digest”** authorization mode is selected. The default user name is **“Admin”**.
- **IP01-IP05:** enter the IP address of the third party devices when the **“WhiteList”** authorization is selected for the integration.

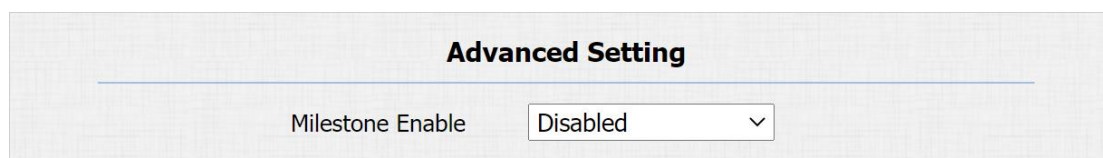
**Please refer to the following description for the Authentication mode**

NO.	Authorization Mode	Description
1	<b>None</b>	No authentication is required for HTTP API as it is only used for demo testing.
2	<b>Normal</b>	This mode is used by Akuvox developer only.
3	<b>WhiteList</b>	If this mode is selected, you are only required to fill in the IP address of the third party device for the authentication. The whitelist is suitable for operation in the LAN.
4	<b>Basic</b>	If this mode is selected, you are required to fill in the User name and the password for the authentication. In Authorization field of HTTP request header, use Base64 encode method to encode of username and password.
5	<b>Digest</b>	Password encryption method, only supports MD5. MD5( Message-Digest Algorithm) In Authorization field of Http request header: WWW-Authenticate:Digest realm="HTTPAPI",qop="auth,auth-int",nonce="xx", opaque="xx".
6	<b>Token</b>	This mode is used by Akuvox developer only.



## 22.3.Integration with Milestone

Akuvox multi-tenant door phone integrate with Milestone surveillance system, to do this configuration on the web **Intercom > Onvif** interface.

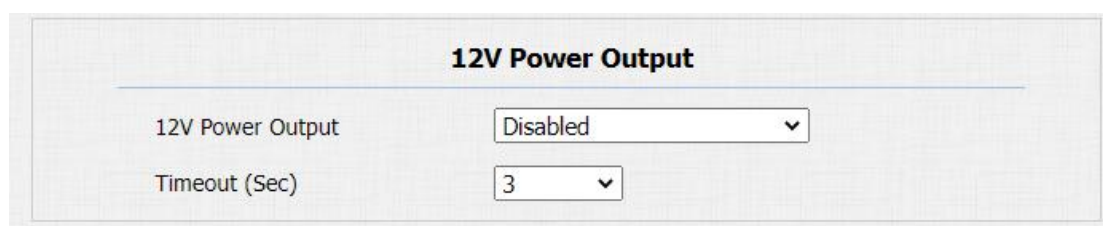


### Parameter Set-up:

- **Milestone Enable:** enable to integrate with Milestone system. It is disabled by default.

## 22.4.Power Output Control

R28 can serve as a power supply for the external relays. Path: **Intercom > Advanced >12V Power Output**.



### Parameter Set-up:

- **12V Power Output:** select **Disabled** to disable the power output function; select **Always** to enable the access controller to provide continuous power to the third party device. Select **Triggered By Open Relay** if you want the E18 to provide power to the third party device via 12 output and GND interface during the timeout when the status of relays is shifted from low to high.

- **Time Out (Sec):** select the power supply time duration after the relay is triggered. Three options: 3, 5, 10. It is 3 seconds by default. The power output is 12V , and the maximum output amperage is 0.8A.

## 22.5.Integration with Lift Control

Akuvox multi-tenant door phone integrated with ZKTECO lift control. The local contact, card, and privatekey of the device can be edited by editing the corresponding floor field to recruit elevators at the same time as the door is opened. If the device connects Akuvox SmartPlus, and the cloud sends the corresponding floor information to open the door. Bluetooth, PIN code, App both can open the door at the same time. The device connects to the cloud mode and can only be placed in the public equipment under the building. At this time, Unit-APT can be used to match the corresponding floor. For example: 2-803, where 803 is the room number, the last two 03 represents the room number, and 8 represents the floor number. To do this configuration on the web **Intercom > Lift Control** interface.

Lift Control	
Lift Control	Disabled ▾
Server IP	<input type="text"/>
Server Port	<input type="text"/>
Timeout	60 (1~60s)

### Parameter Set-up:

- **Lift Control:** enable to use lift control feature.
- **Server IP:** enter the ZKT lift control panel IP address.
- **Server Port:** enter the lift control port of the IP address.
- **Timeout:** The default value is 60, the value range is 1~60s, which means that within the Timeout value after the door is opened successfully, the corresponding floor button can be pressed. After timeout, it will not take effect.

## 23. Password Modification

### 23.1. Modifying Device Web Interface Password

To change the default web password on web **Security > Basic** interface. Select "**admin**" for the administrator account and "**User**" for the User Account. Click the **Change Password** tab to change the password. You can also disable User account permission to login in the web.

**Web Password Modify**

---

User Name admin ▾ Change Password

---

**Account Status**

---

Admin Enabled ▾

User Disabled ▾

### 23.2. Modifying Device System Password

You can modify both device system password and service (setting) password on the web or on the device. On the web, navigate to **Security > Basic > LCD Password Modify**, and change the passwords if needed. The default system password is \*2396#, the default **service (setting)** password is 3888.

**LCD Password Modify**

---

System Password  (4 Digits) Apply

Service Password  (4 Digits) Apply

To set or modify the system and service passwords on the device, press the system password ( which is \*2396# by default), select "**Admin Access**", and select "**Set Admin Password**", then set the system password. Select "Set Service Password", then set the service password. You can delete the passwords if needed.

## 23.3. Configure Web Interface Automatic Logout

It is a protection design. When there is no operation on the website and when the Session Time Out Value time is reached, the website will automatically log out. To do the configuration on the web **Security > Basic > Session Time Out** interface.

Session Time Out	
Session Time Out Value	<input type="text"/> (60~14400s)

### Parameters Set-up:

- **Session Time Out Value:** the range from 60 to 14400 sec. If there is no operation over time, you need to log in the website again.

## 24. System Reboot&Reset

### 24.1.Reboot

If you want to restart the device system, you can operate it on the device **Upgrade > Basic** web interface as well.

### 24.2.Reset

If you want to reset the device system to the factory setting, you can it on the web **Upgrade > Basic** interface.

## 25. Abbreviations

**ACS:** Auto Configuration Server

**Auto:** Automatically

**AEC:** Configurable Acoustic and Line Echo Cancelers

**ACD:** Automatic Call Distribution

**Autop:** Automatical Provisioning

**AES:** Advanced Encryption Standard

**BLF:** Busy Lamp Field

**COM:** Common

**CPE:** Customer Premise Equipment

**CWMP:** CPE WAN Management Protocol

**DTMF:** Dual Tone Multi-Frequency

**DHCP:** Dynamic Host Configuration Protocol

**DNS:** Domain Name System

**DND:** Do Not Disturb

**DNS-SRV:** Service record in the Domain Name System

**FTP:** File Transfer Protocol

**GND:** Ground

**HTTP:** Hypertext Transfer Protocol

**HTTPS:** Hypertext Transfer Protocol Secure Socket Layer

**IP:** Internet Protocol

**ID:** Identification

**IR:** Infrared

**LCD:** Liquid Crystal Display

**LED:** Light Emitting Diode

**MAX:** Maximum

**POE:** Power Over Ethernet

**PCMA:** Pulse Code Modulation A-Law

**PCMU:** Pulse Code Modulation  $\mu$ -Law

**PCAP:** Packet Capture

**PNP:** Plug and Play

**RFID:** Radio Frequency Identification

**RTP:** Real-time Transport Protocol

**RTSP:** Real Time Streaming Protocol

**MPEG:** Moving Picture Experts Group

**MWI:** Message Waiting Indicator

**NO:** Normal Opened

**NC:** Normal Connected

**NTP:** Network Time Protocol

**NAT:** Network Address Translation

**NVR:** Network Video Recorder

**ONVIF:** Open Network Video Interface Forum

**SIP:** Session Initiation Protocol

**SNMP:** Simple Network Management Protocol

**STUN:** Session Traversal Utilities for NAT

**SMTP:** Simple Mail Transfer Protocol

**SDMC:** SIP Devices Management Center

**TR069:** Technical Report069

**TCP:** Transmission Control Protocol

**TLS:** Transport Layer Security

**TFTP:** Trivial File Transfer Protocol

**UDP:** User Datagram Protocol

**URL:** Uniform Resource Locator

**VLAN:** Virtual Local Area Network

**WG:** Wiegand

## 26. FAQ

Q1: How to obtain the IP address of R2X

A1: ✓ For devices with a single button - E21/ R20/ R23/ R26:

While E21/ R20/ R23/ R26 power up normally, hold the call button for 5 seconds after the statue LED turns blue and it will enter into IP announcement mode. In announcement mode, the IP address will be announced repeatedly. Press call button again to quit the announcement mode.

✓ For devices with multiple numeric keyboard - R27:

While R27 power up normally, press "\*2396#" to enter the home screen and press "1" to go to system Information screen to check the IP address.

✓ For devices with touch screen - X915/R29:

While it power up normally, in the dial interface, press "9999", "Dial key", "3888" and "OK" to enter the system setting screen. Go to info screen to check the IP address.

✓ Common method:

Using Akuvox IP Scanner to search Akuvox devices in the same LAN network.

Q2: Do Akuvox devices support opus codec?

A2: For now, only Akuvox Android video IP phone R48G can support Opus audio codec.

Q3: What is the supported temperature range for akuvox doorphone?

A3: R20/E21/R26/R23/Standard R27/Standard X915 -- 14° to 112°F (-10° to 45°C)

R27/X915 with heating supporting --- 40 degrees

R28 -- (-40°C~55°C)

Indoorphone -- 14° to 112°F (-10° to 45°C)

IPPhone -- 32°~104°F(0~40°C)

Q4: Do Akuvox devices support Modbus protocol?

A4: No.

Q5 : Failure in importing the X915 face data to another X915 using the exported face data.

A5: Please confirm the following steps:

The import format is zip;

1. After you export, you need to unzip the .tgz folder, then make the unzipped folder into .zip again.

Q6: Which version of ONVIF does R20 and X915 support?



A6: Onvif 18.04 profiles

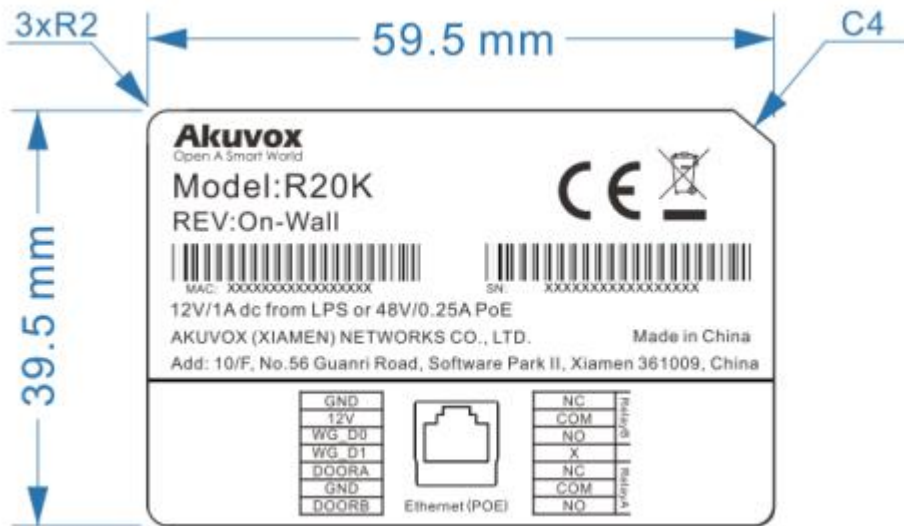
Q7: Do door phones support these card types? Prox, Legacy iClass,iClassSE,HID Mifare, HID DESFire,HID SEOS

A7: Sorry, they are not supported. They need to be implemented via hardware modifications.

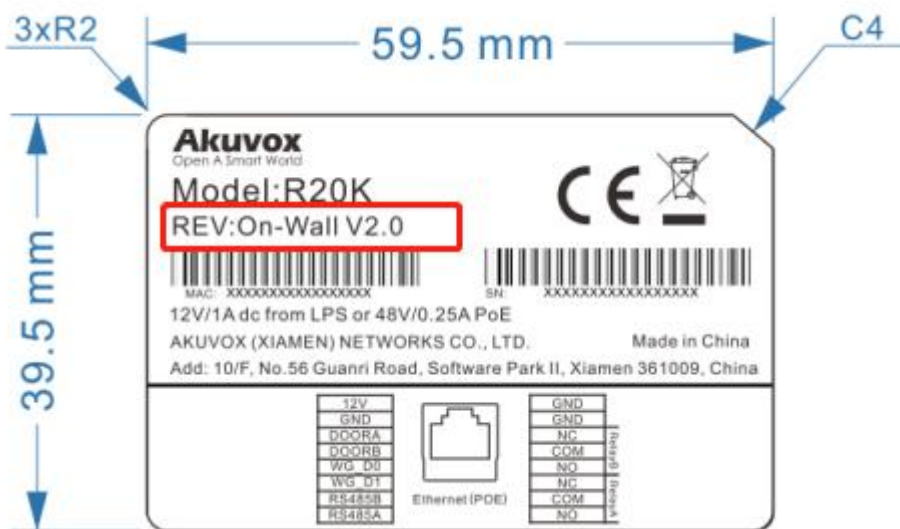
Q8: How to confirm whether my device is hardware version 1 or hardware version 2?

A8: 1.Label

● **Hardware version 1**



● **Hardware version 2**



- **Firmware Version**

The firmware is different between hardware version1 and hardware version 2.

Go to Web-Status -Firmware Version.

20.X.X.X is hardware version 1.

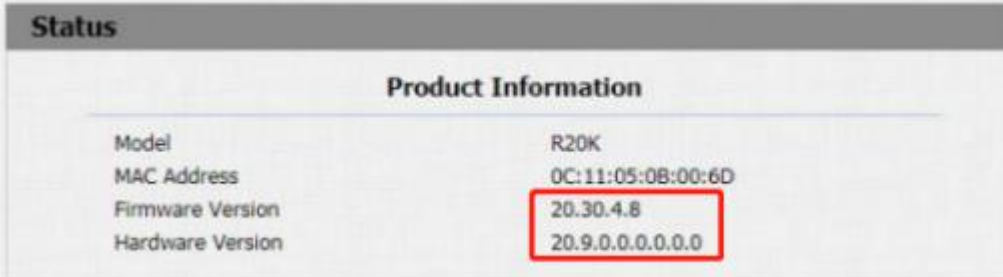
220.X.X.X is hardware version 2.

- **Hardware version**

The firmware is different between hardware version1 and hardware version 2.

Go to Web-Status -Firmware Version.

If the hardware version is 220.x, then the device is hardware version 2.



Status	
Product Information	
Model	R20K
MAC Address	0C:11:05:0B:00:6D
Firmware Version	20.30.4.8
Hardware Version	20.9.0.0.0.0.0

## 27. Contact us

For more information about the product, please visit us at [www.akuvox.com](http://www.akuvox.com) or feel free to contact us by

Sales email: [sales@akuvox.com](mailto:sales@akuvox.com)

Technical support email: [support@akuvox.com](mailto:support@akuvox.com)

Telephone: +86-592-2133061 ext.7694/8162

We highly appreciate your feedback about our products.

