

Basic SIP Paging Adapter

User Manual



VOXPRIME

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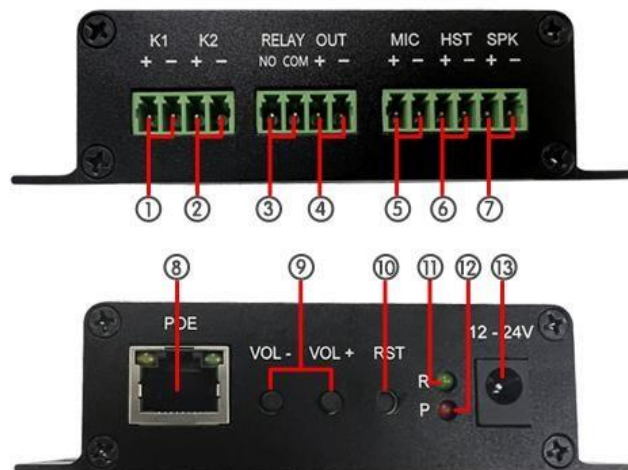
1. Overview

SIP-T20 is a IP based paging adapter that can covert analog to SIP. It's small and portable design with black outlook. The various interfaces (MIC, headset and speaker) make it possible for quickly configure intercom and paging solution. It's compatible with SIP & ONVIF protocol that can be used in VoIP and security field. It supports two-way intercom communication. Flexible Alarm in and out solutions(GPIO, HTTP URL, and relay out) are widely applied in daily life. The 48K OPUS Audio Codec enables excellent sound quality to make announcement, play background music, security alarm in school, factory and hospital, etc.





2. Interface Description



<p>① K1 DSS Key</p>	<p>Connect to:</p> <ol style="list-style-type: none"> external keys infrared probe and emergency switch door sensor and other switch components
<p>② K2 DSS Key</p>	<p>Connect to:</p> <ol style="list-style-type: none"> external keys infrared probe and emergency switch door sensor and other switch components
<p>③ Relay NO COM Interface</p>	<p>Control the alarm system on/off</p>

④ IO Onput Interface	<p>Responding to:</p> <ol style="list-style-type: none"> 1. control the external amplifier power switch 2. the short-circuit input interface 3. login device security page settings 4. control the alarm light, electric locks and other equipment 5. with the adjacent power port connection for external equipment
⑤ Microphone Interface	2.2K Ohm impedance electric condenser microphone is recommended.
⑥ Headset Interface	Speaker audio line signal output impedance 32 Ohm, single ended output voltage 1.2V, used for external headphones or amplifier.
⑦ Speaker Interface	Maximum support 15W speaker.
⑧ Ethernet Interface	WAN port, standard RJ45 interface, 10/ 100M adaptive, support POE input.
⑨ Volume Control Key	This two keys is to adjust the volume of the device's, bell, phone call and broadcasting, etc.
⑩ System Reset Key	Press Rest key and hold for 3 seconds, the devices will restart to factory setting.
⑪ Run Indicator	The light is on shows that the device is working well.
⑫ Power Indicator	The light on shows that the power is connected.
⑬ Power Input Interface	12V ~ 24V 2A input, according to the input voltage to determine the maximum output power amplifier.

3. Web Configuration

Web configuration includes complete function setting. When the device and your computer are connected to a same network, please open a browser and type in <http://192.168.5.200>, then log in with defaulted username and password as below.

Username: admin

Password: tm1234

Login IP Speaker

Username

Password

[Forgot Password?](#)

3.1 Status

You can check out firmware version, free space and two SIP accounts status of SIPT20, also can locate the current network information here, like MAC, IP address and gateway etc.

- Status
- Basic
- ONVIF
- SIP Account
- Audio
- Media File
- Alarm
- Http URL
- Schedule
- RTP Multicast
- Firewall
- System

Status

Device Time	2022-03-01 09:02:31
Serial Number	50346849A878571C
Firmware Ver	T20-V3.1.2
Free Space	184KB
SIP1 Status	NONE
SIP2 Status	NONE

Network

MAC Address	A2:C0:A4:75:3B:99
IP Address	192.168.5.200
Subnet Mask	255.255.255.0
Gateway	192.168.5.1
Primary DNS	192.168.5.1
Secondary DNS	218.85.152.99

3.2 Basic

3.2.1 Date/ Time

There are two update modes for time: NTP/ local time, choose one and set the time zones, NTP sever and interval can choose default setting, then save the configuration.

The image displays two screenshots of the 'Date/Time' configuration interface. Both screenshots have a green header bar with the text 'Date/Time'.

The top screenshot shows the NTP configuration mode. It includes the following fields:

- Device Time: 2022-04-28 13:47:36
- Update Mode: NTP (dropdown menu)
- TimeZone: GMT+08:00 (dropdown menu)
- NTP Server: pool.ntp.org (dropdown menu)
- NTP Interval: 10 (text input) Minutes
- Save button (green)

The bottom screenshot shows the LocalTime configuration mode. It includes the following fields:

- Device Time: 2022-04-28 13:47:36
- Update Mode: LocalTime (dropdown menu)
- LocalTime: 2022-04-28 14:04:02 (text input)
- Save button (green)

3.2.2 Network

When you choose DHCP and save it, IP address will be created automatically by a DHCP server, then you need to login again with the new IP address on browser: 192.168.5.XXX.

Status IP address: it is a default IP and will not be changed as following.

Network

DHCP
 Static IP Address

IP Address:
 Subnet Mask:
 Gateway:
 Primary DNS:
 Secondary DNS:

3.3 ONVIF

Select Enable ONVIF, then the device be searched by ONVIF VMS.

Default user name: admin, password:tm1234.

Status

Basic

ONVIF

SIP Account

Audio

Media File

ONVIF

ONVIF Enable

User Name:

Password:

3.4 SIP Account

Each device has two SIP accounts, put SIP extension messages into the blanks and save the configuration, then you can check if it registers successfully or not on status.

Expire time	Set the expire time of registered account information
Ringing tone	5 system ringtones and 10 users upload media files
Auto Answer	answer immediately and answer delay when a calling incomes

The screenshot displays the 'SIP Set' configuration page. On the left is a sidebar with the following menu items: Status, Basic, ONVIF, SIP Account (highlighted in green), Audio, Media File, Alarm, Http URL, Schedule, RTP Multicast, Firewall, and System. The main content area is titled 'SIP Set' and contains the following fields:

- Account: Account 1 (dropdown)
- User Name: (text input)
- Password: (text input)
- Display Name: (text input)
- Server Host: (text input)
- Server Port: (text input)
- Expire Time: 3600 (text input) Seconds
- Ringing Tone: bell1 (dropdown) with a play button icon
- Auto Answer: Answer Immediately (dropdown)
- Incoming Notify: * Please set incoming notify at Alarm In
- Answer Notify: * Please set answer notify at Alarm In
- Close Notify: * Please set close notify at Alarm In

A green 'Save' button is positioned at the bottom right of the configuration area.

3.5 Audio

ACE(acoustic echo canceling): to make a perfect sound quality.

Mic / out volume: adjust mic and output volume at 0-100.

Jitter buffer: to make the audio more stable.

Amp auto off: It's set defaulted as ON, then there is no noise when not broadcasting.

Code setting: four audio codes to compatible with major audio sources.

Audio

AEC Enable

Mic volume (0-100)

Out Volume (0-100)

Jitter Buffer (60 - 2000) ms

Amp Auto OFF

Codec Setting

<input checked="" type="checkbox"/>	OPUS
<input checked="" type="checkbox"/>	G.722
<input checked="" type="checkbox"/>	G.711U
<input checked="" type="checkbox"/>	G.711A

3.6 Media File

There are five system ringtones, and you can upload 10 media files as customers' demands: music, announcement, bells, etc.

3.7 Alarm

We can set 2 DSS keys and 2 SIP accounts to realize alarm function, to ready the combination with alarm system.

- Status
- Basic
- ONVIF
- SIP Account
- Audio
- Media File
- Alarm
- Http URL
- Schedule

Alarm In

Input

File Enable

Sip Enable

Uri Enable

Output Enable

Relay Enable

3.7.1 DSS Key Setting

- Enable the file, you select a action type(start/ stop), play file and cycle mode, save the configuration, then press buttonK1 & K2, the bell will ring/close.

Alarm In

Input: Key 1

File Enable:

Action Type: Start

Play File: bell1

Cycle Mode: Once only (dropdown menu open)

Sip Enable:

Url Enable:

Output Enable:

Relay Enable:

Save

- SIP enable: choose a SIP account you register, SIP action: call out/hang up, you can put the SIP number, eg: 8112, make sure it's the extensions which connected to the same IP sever with SIP account 1&2.

If you select call out, and press K1/K2 button, then extension 8112 will receive a call.

Alarm In

Input: Key 1

File Enable:

Sip Enable:

Sip Account: Account 1

Sip Action: Call Out

Sip Number: 8112

- Enable URL: put the HTTP URL, after pressed K1/K2, the URL will be working.

- Output enable: turn on/off the output, press K1/K2, the output succeeds.
- Relay enable: turn on/off the output, press K1/K2, the relay succeed.

Alarm In

Input	<input type="text" value="Key 1"/>
File Enable	<input type="checkbox"/>
Sip Enable	<input type="checkbox"/>
Uri Enable	<input checked="" type="checkbox"/>
Http URL	<input type="text"/>
Output Enable	<input checked="" type="checkbox"/>
Output Action	<input type="text" value="On"/> <input type="text" value="10"/> S
Relay Enable	<input checked="" type="checkbox"/>
Relay Action	<input type="text" value="On"/> <input type="text" value="10"/> S

3.8 HTTP URL

User can control the alarm by HTTP URL:

- (1) Enable the selection;
- (2) Open any browser you have in computer;
- (3) Put the URL as the following examples, enter it.

Status

Basic

ONVIF

SIP Account

Audio

Media File

Alarm

Http URL

Schedule

RTP Multicast

Firewall

System

Http URL

Play File Enable

Example1: `http://192.168.5.200/api/play?action=start&file=bell1`

Example2: `http://192.168.5.200/api/play?action=start&file=userfile1&mode=once&volume=10`

Example3: `http://192.168.5.200/api/play?action=start&file=userfile1&mode=multiple&count=10&volume=20`

Example4: `http://192.168.5.200/api/play?action=start&file=userfile1&mode=duration&count=10&volume=30`

Example5: `http://192.168.5.200/api/play?action=stop`

Sip Call Enable

Example1: `http://192.168.5.200/api/sipcall?action=call&number=100&line=auto`

Example2: `http://192.168.5.200/api/sipcall?action=call&number=100&line=1`

Example3: `http://192.168.5.200/api/sipcall?action=hangup`

Output Enable

Example1: `http://192.168.5.200/api/output?action=on`

Example2: `http://192.168.5.200/api/output?action=on&duration=10`

Example3: `http://192.168.5.200/api/output?action=off`

Relay Enable

Example1: `http://192.168.5.200/api/relay?action=on`

Example2: `http://192.168.5.200/api/relay?action=on&duration=10`

Example3: `http://192.168.5.200/api/relay?action=off`

3.9 Schedule

This function is widely use in school, factory and office projects. Making a regular bell, announcement and alarm.

Enable the schedule, you can name the schedule. then setting it step by step.

Schedule Add/Edit

Schedule Enable

Schedule Name

Start Date 2022/01/01

End Date 2099/12/31

Allowed Days Mon Tue Wed Thu Fri Sat Sun

Action Time 08:00

Action Type

Play File

Cycle Mode

Times (1-1000)

Duration (1-60000) Seconds

3.10 RTP Multicast IP

There are 10 RTP addresses can be received for each device, please note that: port numbers do not use continuous numbers when setting the same RTP addresses. Use discontinuous numbers. eg:

239.255.1.2:8000, 239.255.0.1:8001, 239.255.0.1:8002 (×)

239.255.0.1:8000, 239.255.0.1:8002, 239.255.0.1:8004 (√)

- Multicast address range: 224.0.0.0-239.255.255.
- Ports range: 1024-65536
- Use IP Tool, Audio manager and PA System to make RTP multicast.

- Status
- Basic
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RTP Multicast

Priority	IP Address (e.g. 239.255.0.1:5004)
1	<input type="text" value="239.255.1.2:8000"/>
2	<input type="text" value="239.255.1.2:8002"/>
3	<input type="text" value="239.255.1.2:8004"/>
4	<input type="text" value="239.255.1.2:8006"/>
5	<input type="text" value="239.255.1.2:8008"/>
6	<input type="text" value="239.255.1.2:8010"/>
7	<input type="text" value="239.255.1.2:8012"/>
8	<input type="text" value="239.255.1.2:8014"/>
9	<input type="text" value="239.255.1.2:8016"/>
10	<input type="text" value="239.255.1.2:8018"/>

3.11 Firewall

This function is use to protect your network safety. You can edit the firewall automatic defence rules as you need as follows.

- Status
- Basic
- ONVIF
- SIP Account
- Audio
- Media File
- Alarm
- Http URL
- Schedule
- RTP Multicast
- Firewall
- System

Firewall Rules

#	Name	Type	IP/MAC	Action
1				
2				
3				
4				
5				

Automatic Defense Rules

#	Name	Protocol	Port Range	Rate
1		-		
2		-		
3		-		
4		-		
5		-		

3.12 System

3.12.1 Upgrade

When reboot and reset the system, system will turn to original setting, and you need to re-login the web page.

How to upgrade SIP-T20 firmware version in web interface?

- (1) Select the latest version firmware T20-xxx-bin.
- (2) Click upgrade to refresh, it would require about 20s.
- (3) Re-login the web interface, latest version has upgraded.

3.12.2 Security

Set a new user name and password as you need, save the configuration and restart login.

The screenshot displays the web interface configuration page. On the left is a sidebar menu with options: Status, Basic, ONVIF, SIP Account, Audio, Media File, Alarm, Http URL, Schedule, RTP Multicast, Firewall, and System (highlighted in green). The main content area is divided into two sections:

- Upgrade:** Contains three buttons: 'Reboot' (with text 'Reboot Device Now'), 'Reset' (with text 'Reset to Factory Setting'), and 'Upgrade' (with a file selection button labeled '选择文件' and the text '未选择任何文件').
- Security:** Contains five input fields: 'User Name' (pre-filled with 'admin'), 'Password' (masked with '*****'), 'New User Name', 'New Password', and 'Confirm Password'. A green 'Save' button is located at the bottom right of this section.

4. IPTool Configuration

Apart from Web configuration, IPTool is the other option that configure quickly basic information such as SIP account setting, volume setting, RTP Multicast setting, upgrade. Please follow below steps.

- (1) Download IPTool in <https://www.tonmind.com/category/downloads/5>
- (2) Enter IPTool, scan local network, the device will appear and then start setting.

IPToolCS20

Network Search RTP Multicast Settings

No.	UID	Name	MAC	IP address	Netmask	Gateway	SIP Settings	RTP Settings	Version	Audio	System
<input type="checkbox"/> 1	5023428330897A1C	CS20	a2c0a426a0ce	192.168.5.192	255.255.255.0	192.168.5.1		239.255.0.0:5008	CS20-V2.8.1N	12,7	admin

Current Device CS20-5023428330897A1C

SIP Account 1

User Name

Password

Display Name

Server Host

Server Port

SIP Account 2

User Name

Password

Display Name

Server Host

Server Port