



Paging Microphone | Quick Start Guide

Step 1: Interface Specifications

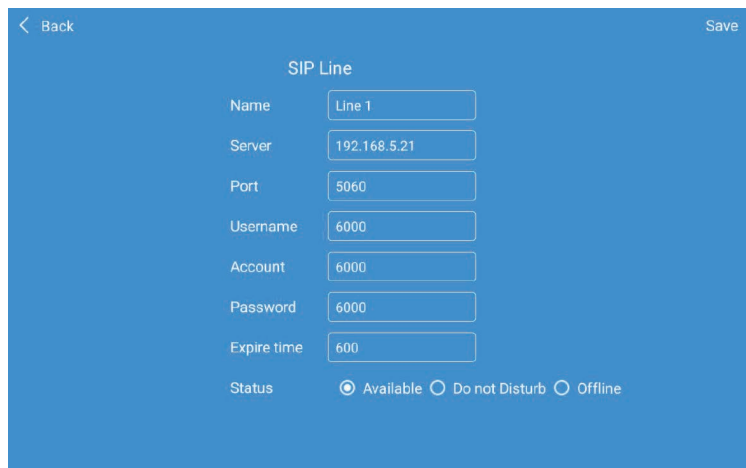
1. Volume Switch for microphone, range 1-100
2. Volume Switch for background music, range 1-100
3. Volume Switch for speaker, range 1-100
4. Power button for screen display
5. Built-in Microphone



1. 2 audio input and 1 audio output
2. Volume switch for audio output
3. 2 GPIO in and 1 GPIO out(optional choice)
4. RJ45 network port
5. Reset system and firmware button
6. USB 2.0 interface
7. Interface for system upgrading
8. Power interface 12V/2A

Step 2: Application and Setting

2.1 SIP Setting.

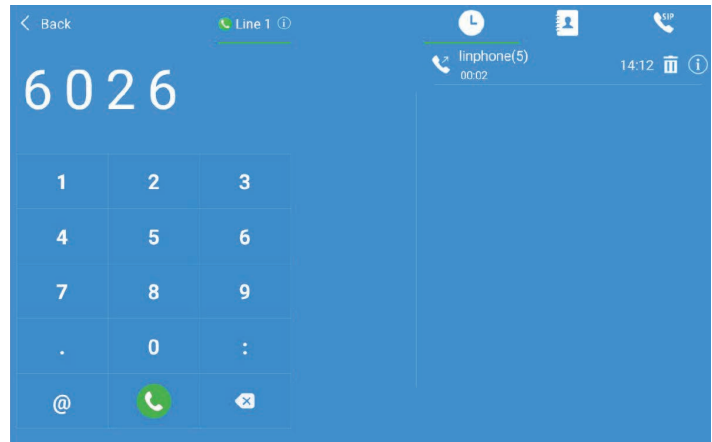


Screenshot of the SIP Line configuration interface. The interface has a blue background and includes a 'Back' button at the top left and a 'Save' button at the top right. The configuration fields are as follows:

SIP Line	
Name	Line 1
Server	192.168.5.21
Port	5060
Username	6000
Account	6000
Password	6000
Expire time	600
Status	☒ Available ☐ Do not Disturb ☐ Offline

2.2 How to Make SIP Call.

1. Click the Green Phone or pick a sip line
2. Extension number can be called by 6026.
3. Make a phone call by P2P eg.
1@192.168.5.200.2 GPIO in and 1 GPIO out(optional choice)



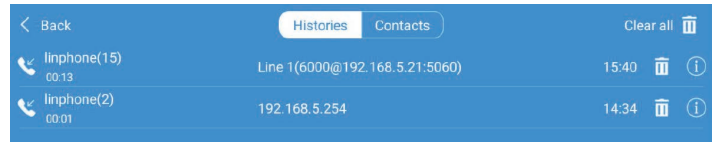
2.3 RTP Multicast

1. Add a RTP address
2. Select audio codec and duration time,
3. Choose paging source (mic/ media/ AUX in),
4. Select a paging mode.



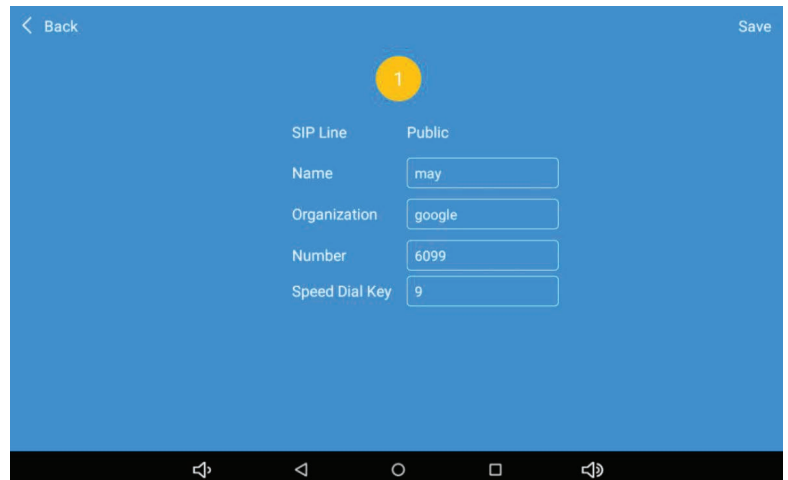
2.4 History & Contact

1. Click the green Clock on the main menu. Calling history & add contact information can be found here.



2.5 History & Contact

2. Add a new contact.
3. Setting the name, organization, numb and speed dial key.



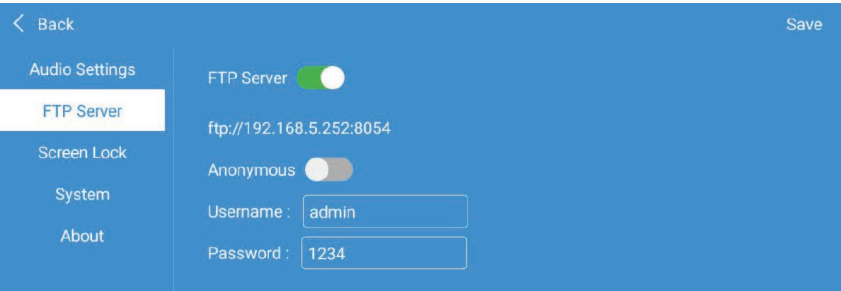
2.5 System Settings

Can set sip audio codec and microphone as follows.

1. Upload music or any MP3 files to FTP sever:
ftp://192.168.5.252:8054(it's a dynamic IP).

2. Use USB to download music.

3. Upload files by FTP tool, like File Zilla.



2.5 Screen Lock Settings

Can set sip audio codec and microphone as follows.

