



AEi COMMUNICATIONS VM-2208-S Full Feature SIP Telephone User Manual

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AEi COMMUNICATIONS VM-2208-S Full Feature SIP Telephone



Product Information

The SIP Phone is a full-feature telephone that provides voice communication. It comes in various models: VM-2208-S(S), VM-2108-S(S), VM-2208-SGA(S), VM-2108-SGA(S), VM-9208-S(S), VM-9108-S(S), VM-9208-SGA(S), and VM-9108-SGA(S). This product is designed to be used with a single Ethernet connection.

Overview

This phone is a full-feature telephone that provides voice communication. Read this QIG carefully to learn how to operate this product and take advantage of its features.

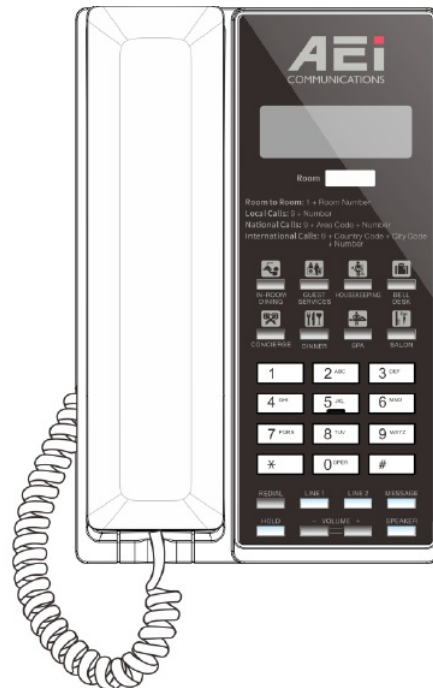
Package Contents

- Base Unit
- Handset
- Stand
- AC Adaptor
- RJ-45 Cable
- Quick Installation Guide (This document)

Product Description

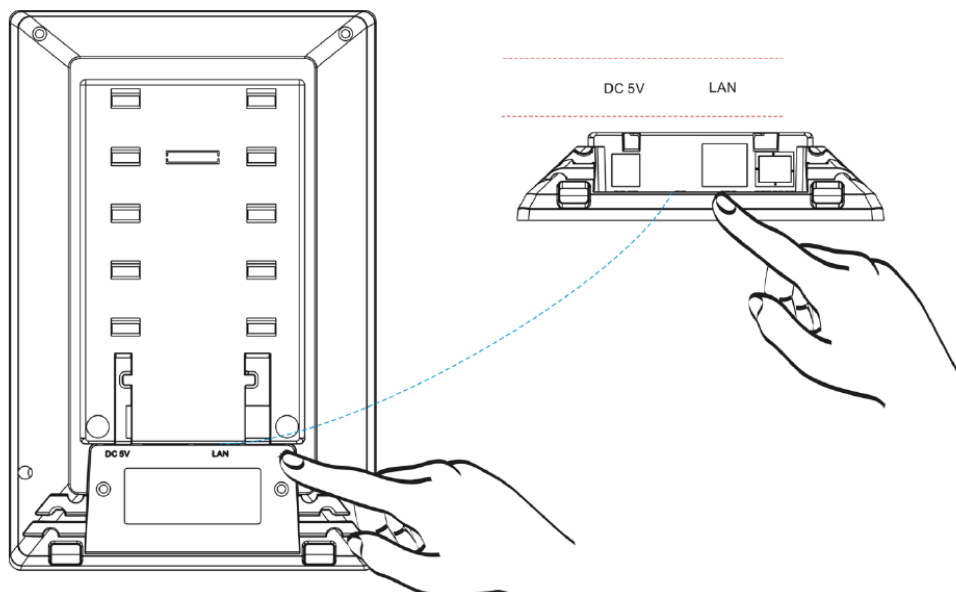
The SIP Phone features a display screen for call and status information, programmable service keys, alphanumerical keys for entering phone numbers and texts, function keys for various call actions, and volume control keys. It also has LAN and DC 5V power ports at the back.

Front View



- Base unit keys:
 - Display screen: To display calls and status information.
 - Programmable services keys: Press to call preset services.
- Alphanumerical keys: Press these keys to enter phone numbers and texts.
- Function keys:
 - LINE 1 / LINE 2: Indicates the line status.
 - HOLD (2 lines only): Puts an active call on hold.
 - MUTE (single line only): Mutes the microphone.
 - REDIAL: Calls the last dialed number.
 - SPEAKER: Activates the speakerphone for hands-free conversations.
 - VOLUME UP/DOWN: Increases or decreases the volume of the ringer tone or speakerphone.
 - MESSAGE: Retrieves voice messages.

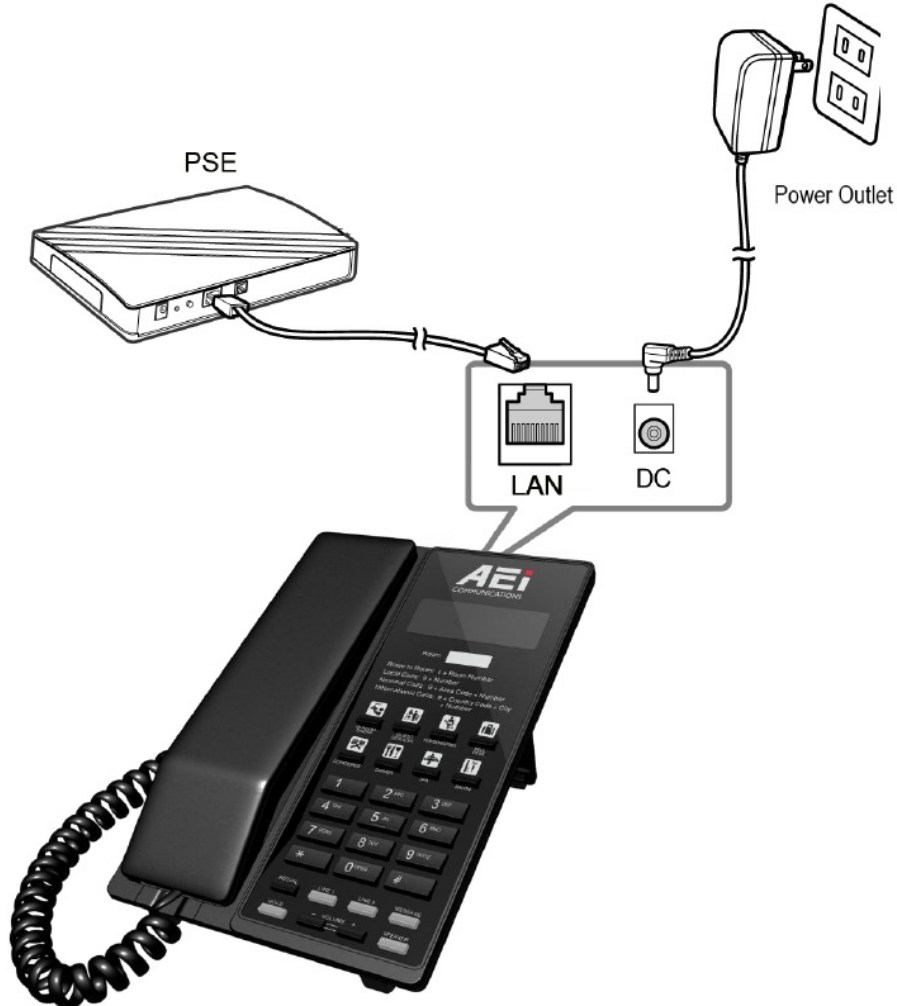
Back View



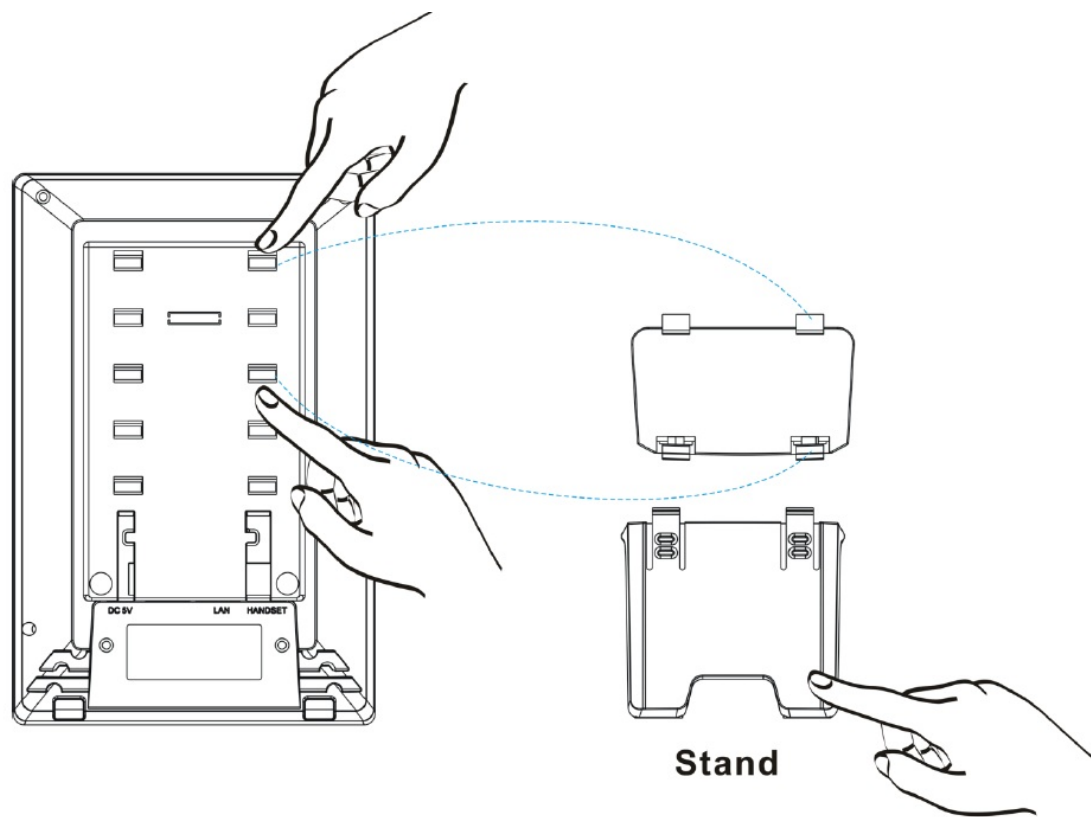
- LAN port: Connects to HUB, ADSL, or Switch(PSE).
- DC 5V Power jack: Connects to AC power adapter (optional if not powered via a PSE).
- Handset jack: Connects to the handset.

Connecting the Phone

The following illustration shows how to connect the power and LAN ports of VoIP phone.



The following illustration shows how to install the phone and stand.



To connect the SIP Phone, follow these steps:

1. Connect the LAN port of the phone to a HUB, ADSL, or Switch(PSE).
2. If required, connect the DC 5V Power jack to an AC power adapter.
3. Connect the Handset jack to the handset.

Operation

Answering a Call

An incoming call is indicated by a ringing tone and a flashing indicator at the top of the units. You can answer the call using one of the following methods:

- Pick up the Handset to answer the call.
- Press the LINE (LINE1/LINE2) key on the base or handset to answer the call.

Making a Call

To make a call using the SIP Phone, follow these steps:

- **Dialing with the Keypad:**
 - Enter the telephone number using the alphanumeric keypad.
 - After entering the number, press the # key to force dialing or wait for 4 seconds for automatic dialing.
- **Using the Programmable Services keys:**
 - Press the pre-assigned service hotkey to quickly dial a commonly used number.

Call Options

While a call is in progress, you have additional options:

- **Using the Speaker Phone:**

- To enable the speakerphone function, press the SPEAKER key on the phone while talking on the phone.
- To terminate the call, press the SPEAKER key again.

- **Putting a Call on Hold**

You may want to put a call on hold temporarily while doing some work or attending to a second incoming or outgoing call. To put a call on hold, do one of the following:

- Press the HOLD key on the phone to put the call on hold.
- When you receive a second call, press the next available LINE key to attend to the new call, and the first call will be put on hold automatically.

- **Checking Voice Messages**

If the MESSAGE key in red, it indicates that there are one or more new messages in voicemail box.

- Press the MESSAGE key on the phone to retrieve voice messages.

(An external ITSP or PBX must support voice message feature. And you need to set a voice message code before using the voice mail feature.)

Support

To configure through the web browser, do the following:

- Ensure the phone is powered up and connected to a DCHP network.
- Press HOLD/MUTE *112*113# (HOLD for 2-Line / MUTE for 1-Line), the IP address is displayed on the LCD screen.
- Open your browser (such as Internet Explorer, Firefox, etc.) and type in the web address of the phone. For example, if the IP address you obtain in step 4 above is 192.168.1.2, enter the web address: http://192.168.1.2:8000
- When the Web login front page is displayed. Enter the user name and the password and click Login. The administrator's user name and password are "admin" and "1234" respectively.

For technological support, please email to support@aeicommunication.com.

Documents / Resources

