

Quick Setup for Voice over IP Service

The *Quick Setup* page is displayed automatically when you first log on ATA. You can use this page to quickly configure connectivity to your provider's Voice over IP network for your analog phone and Cisco SPA302D handsets.

NOTE Connecting to your service provider's network requires Internet connectivity. With the default network settings, your ATA should have Internet connectivity when you connect a cable from the WAN port of the ATA to a port on your router or broadband network device. For more information, see [Internet Settings, page 31](#).

To open this page: Click **Quick Setup** in the menu bar.

STEP 1 Specify the settings for the phone service to be used by each type of device or line. Follow the requirements and recommendations of your service provider. The options are described below.

Device/line types:

- **Line 1:** The phone service used by an analog (FXS) phone or fax machine that is connected to the PHONE port.
- **PSTN:** The phone service used by a phone line that is connected from the LINE port to the PSTN.
- **DECT Line 1:** The phone service used by all connected Cisco SPA302D cordless handsets (when using the default settings in *DECT Handset Outgoing Line Selection* and *DECT Line Contact List* sections). You can add additional phone services for these handsets on the *Voice > DECT Line 2* to *DECT Line 10* pages.

Settings:

- **Proxy:** Enter the domain name or URL of the service provider's proxy server.
- **Display Name:** Enter the name that you want to use to identify your account. This name typically is used as your Caller ID name.
- **User ID:** Enter the user ID that is required to log in to your Internet account.

- **Password:** Enter the password that is required to log in to your Internet account.
- **Dial Plan** in (*Line* section only): Keep the default settings (recommended) or edit the dial plan to suit your site. For more information, see [Configuring Dial Plans, page 225](#).

STEP 2 DECT Handset Outgoing Line Selection: For each DECT Handset, check the boxes to choose the DECT Lines for outgoing calls. Uncheck the boxes for the lines that you do not want to use.

- If you are using only one phone service for all Cisco SPA302D handsets, simply configure the DECT Line 1 settings above and keep the default settings in this section.
- If you have multiple lines, you can select multiple lines for each handset. Alternatively, check the **All Lines** box to make all lines. The enabled options will be listed on the phone screen when the user displays the call options or holds down the green call button.
- Choose a **Default** line, which will be selected automatically for a call when the user presses the green call button.
- Optionally, if you enabled multiple lines, enable **Failover** by selecting **yes**. When this feature is enabled and a call fails through the selected line, the ATA automatically attempts to place the call over another enabled DECT line.

NOTE Cisco SPA232D now supports PSTN to DECT and DECT to PSTN outgoing line failover.

STEP 3 DECT Line Contact List: For each line, check the boxes to choose the handsets that ring when an incoming call is received. Uncheck the boxes for the handsets that you do not want to ring. Check the **All Handsets** box to ring all handsets for the specified line.

STEP 4 Click **Submit** to save your settings. The voice service will restart.

STEP 5 To verify your progress, perform the following tasks:

- a. Click **Voice** in the menu bar, and then click **Info** in the navigation tree. Verify that the Registration State is Registered for all configured lines (Line 1 Status, PSTN Line Status, and DECT 1 Status ~ DECT 10 Status).

If the line is not registered, you may need to refresh the browser several times because it can take a few seconds for the registration to complete. Also verify that your Internet Settings, including DNS server settings, are configured according to the information from your ISP. For more information, see [Internet Settings, page 31](#).

- b. Use an external phone to place an inbound call to the telephone number that was assigned by your ITSP. Verify that the phone rings and you have two-way audio on the call.
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