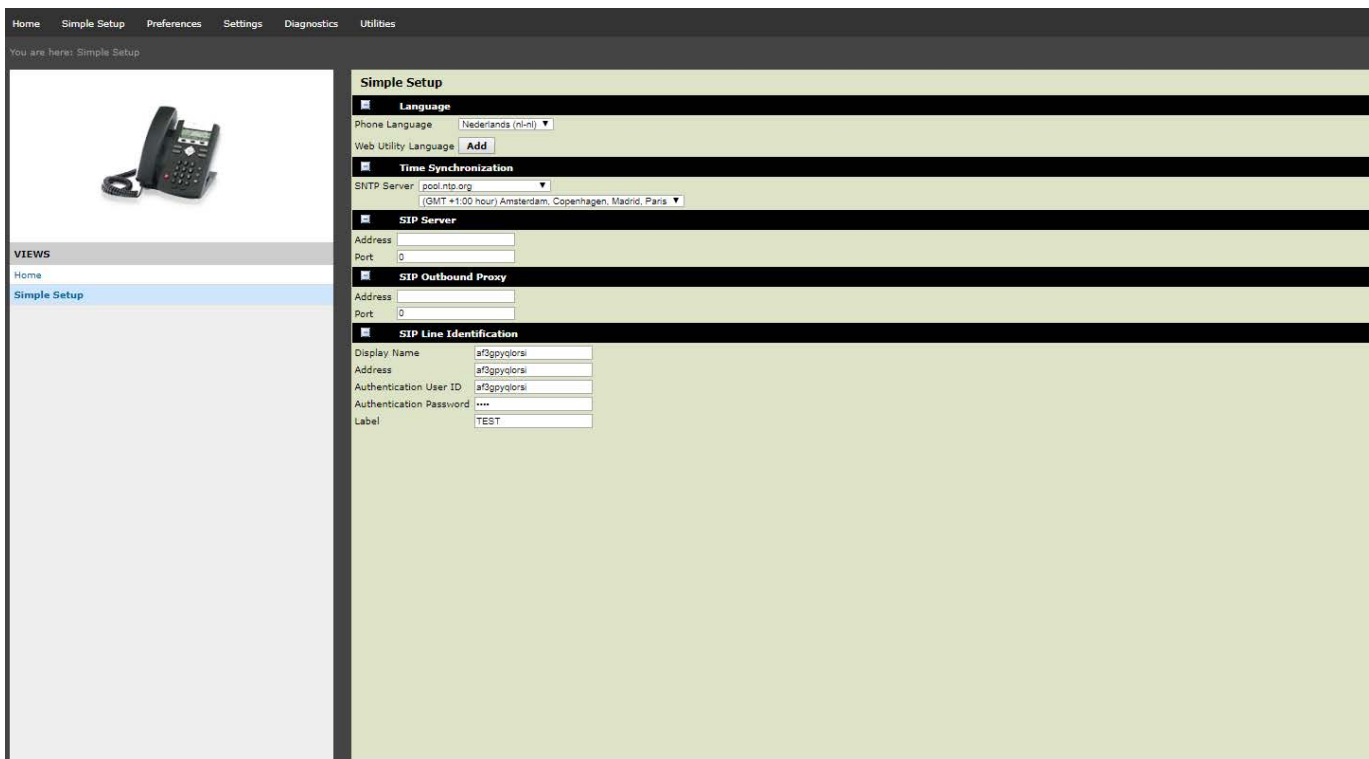


Login als Admin op webinterface met Password: 456



Home Simple Setup Preferences Settings Diagnostics Utilities

You are here: Simple Setup

Simple Setup

Language
Phone Language: Nederlands (nl-nl)
Web Utility Language: Add

Time Synchronization
SNTP Server: pool.ntp.org
(GMT +1:00 hour) Amsterdam, Copenhagen, Madrid, Paris

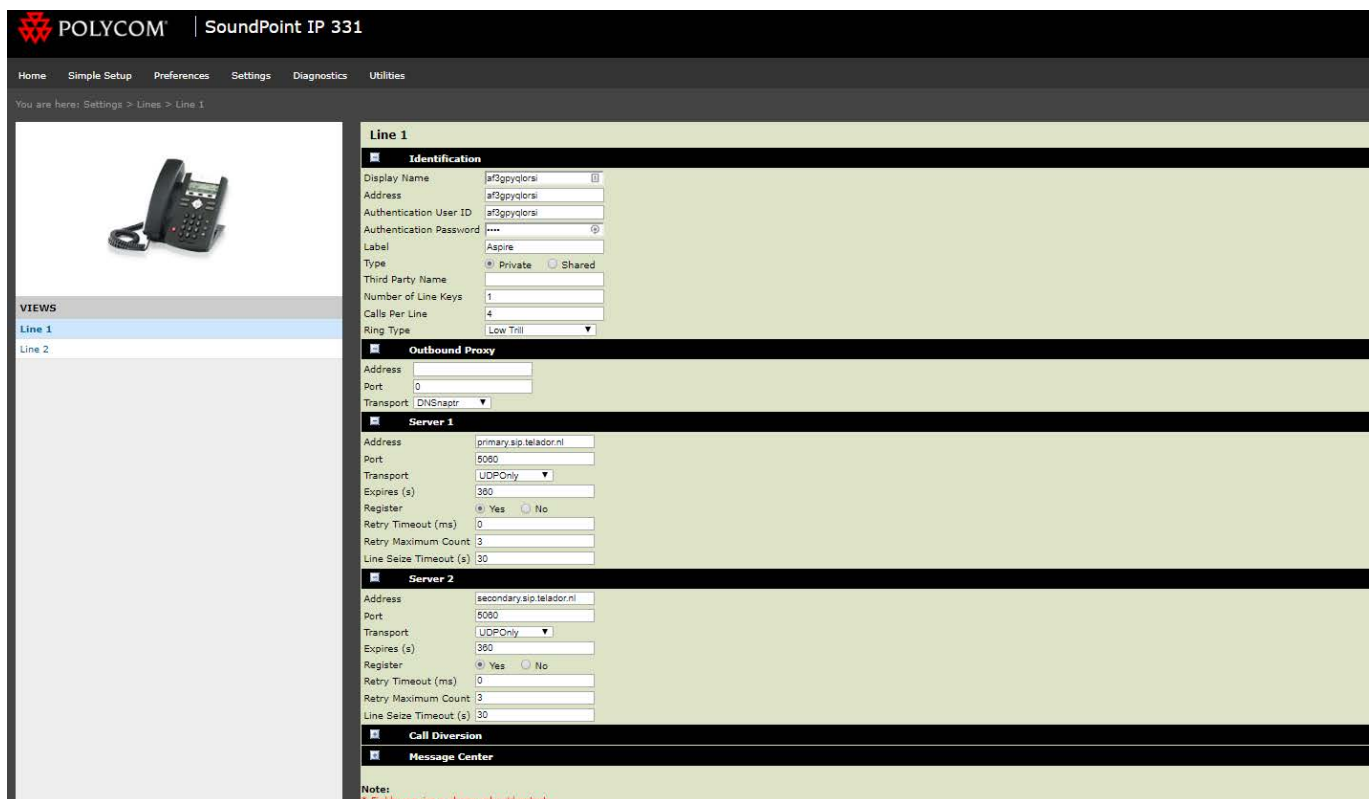
SIP Server
Address:
Port: 0


SIP Outbound Proxy
Address:
Port: 0

SIP Line Identification
Display Name: af3gpyclorsi
Address: af3gpyclorsi
Authentication User ID: af3gpyclorsi
Authentication Password: ****
Label: TEST

Vul bij Display name, address, en Authentication User ID de SIP gegevens uit de portal in
Vul bij Authentication Password het wachtwoord van het SIP account uit de portal in.

Bij "Label" kan een naam worden ingevuld, die in het scherm wordt getoond



 POLYCOM | SoundPoint IP 331

Home Simple Setup Preferences Settings Diagnostics Utilities

You are here: Settings > Lines > Line 1

Line 1

Identification
Display Name: af3gpyclorsi
Address: af3gpyclorsi
Authentication User ID: af3gpyclorsi
Authentication Password: ****
Label: Aspire
Type: Private Shared
Third Party Name:
Number of Line Keys: 1
Calls Per Line: 4
Ring Type: Low Trill

Outbound Proxy
Address:
Port: 0
Transport: DNSnatr

Server 1
Address: primary.sip.telador.nl
Port: 5060
Transport: UDPOnly
Expires (s): 360
Register: Yes No
Retry Timeout (ms): 0
Retry Maximum Count: 3
Line Seize Timeout (s): 30

Server 2
Address: secondary.sip.telador.nl
Port: 5060
Transport: UDPOnly
Expires (s): 360
Register: Yes No
Retry Timeout (ms): 0
Retry Maximum Count: 3
Line Seize Timeout (s): 30

Call Diversion

Message Center

Note:
* Fields require a phone reboot/restart.

Ga naar het tabblad Settings en kies SIP

Pas aan:

[2-9]11|0T|011xxx.T|[0-1][2-9]xxxxxxxx|[2-9]xxxxxxxx|[2-9]xxxT
3|3|3|3|3|3

Vervangen door:

x.T|*x.T|#x.T
3|3|3

The screenshot shows the Polycom SoundPoint IP 331 web interface. The top navigation bar includes 'Home', 'Simple Setup', 'Preferences', 'Settings', 'Diagnostics', and 'Utilities'. The 'Settings' menu is expanded to show 'SIP'. The main content area is titled 'SIP' and contains the following settings:

- Local Settings**
 - Local SIP Port: 5050
 - Calls Per Line Key: 4
 - New SDP Type: Enable Disable
 - Live Communication Server Support: Enable Disable
 - Non Standard Line Seize: Enable Disable
 - Digitmap: x.T|*x.T|#x.T
 - Digitmap Timeout: 3|3|3
 - Remove End-of-Dial Marker: Enable Disable
 - Digitmap Impossible Match: 0
- Outbound Proxy**
- Server 1**
- Server 2**

A note at the bottom states: "Note: * Fields requires a phone reboot/restart." The interface also includes a sidebar with 'VIEWS' such as Microbrowser, Logging, Applications, Audio Codec Priority, Provisioning Server, Syslog, and Paging/PTT Configuration. At the bottom, there are buttons for 'Cancel', 'Reset to Default', 'View Modifications', and 'Save'.

This is a close-up of the SIP settings form from the screenshot. It shows the following fields:

- Digitmap**: A text input field containing the value "x.T|*x.T|#x.T".
- Digitmap Timeout**: A text input field containing the value "3|3|3".
- Remove End-of-Dial Marker**: A radio button selection with "Enable" selected and "Disable" unselected.
- Digitmap Impossible Match**: A text input field containing the value "0".