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Registration and Peering explained

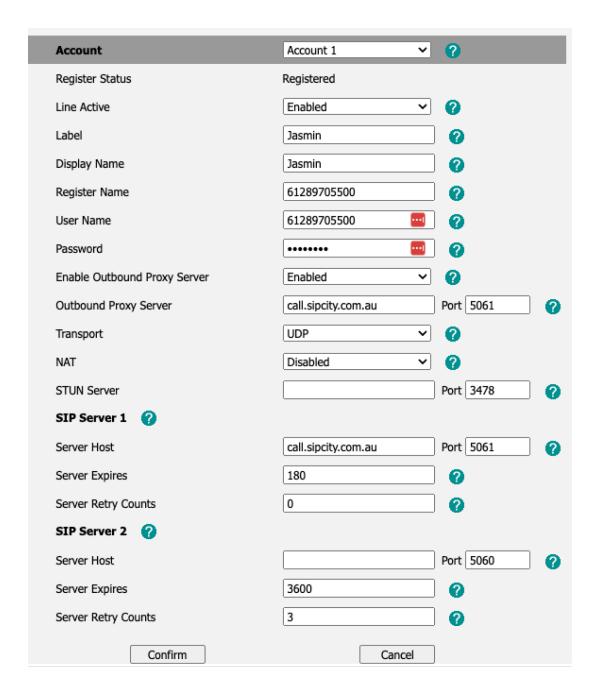
Mike Johnstone - 2024-04-08 - Handsets

Registration and Peering explained

We support most SIP-based VoIP devices including Analog Telephone Adaptors (ATA), IP Phones, IP PBX systems, and Softphones.

Generic SIP Registration

- Display Name: your caller ID
- Proxy Server: plus.sipcity.com
- User name / Account ID / Authenticate ID: (the phone number you are trying to register 1xxxxxxxxxxxx)
- Password: your number's password
- **Transport:** UDP 5060 / TLS 5061
- Server Expires: 180
- Register with the Proxy: yes
- Make calls without registration: no
- Codecs: G711a/u, G729



SIP Peering

Enables a direct trusted IP - IP network connection between your IP-PBX and our voice public IP via a static WAN IP address.

Recommended settings

• **Proxy:** 27.111.12.66 (27.111.12.0/24)

• Recommended Firewall rule: 27.111.12.0/24.

• Ports: UDP 5060 or TLS 5061

We support the following two variations of Peering:

- **Global SIP Peering**: where all numbers on an account are directed to a single customer IP address
- **Standalone**: where a customer only wants to direct calls to a single number to a standalone IP address
- **Trunking**: supports customers with on-premise PBX devices that connect via device registration (UserName and Password). This feature allows an administrator to configure a single registration number to their PBX but preserving the CLI of all outgoing numbers.

See also:

- Ports and IPs.
- Setting up SIP Peering.

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