

General

Line Enable: yes no

Share Line Appearance

Share Ext: public private Shared User ID:

Subscription Expires:

NAT Settings

NAT Mapping Enable: no yes NAT Keep Alive Enable: no yes

NAT Keep Alive Msg: NAT Keep Alive Dest:

Network Settings

SIP TOS/DiffServ Value: SIP CoS Value:

RTP TOS/DiffServ Value: RTP CoS Value:

Network Jitter Level: Jitter Buffer Adjustment:

SIP Settings

SIP Transport: SIP Port:

SIP 100REL Enable: no yes EXT SIP Port:

Auth Resync-Reboot: yes no SIP Proxy-Require:

SIP Remote-Party-ID: no yes Referor Bye Delay:

Refer-To Target Contact: no yes Referee Bye Delay:

SIP Debug Option: Refer Target Bye Delay:

Sticky 183: no yes Auth INVITE: no yes

Call Feature Settings

Blind Attn-Xfer Enable: no yes MOH Server:

Message Waiting: no yes Auth Page: no yes

Default Ring: Auth Page Realm:

Conference Bridge URL: Auth Page Password:

Mailbox ID: Voice Mail Server:

State Agent: CFWD Notify Serv: no yes

CFWD Notifier:

Proxy and Registration

Proxy: Use Outbound Proxy: no yes

Outbound Proxy: Use OB Proxy In Dialog: no yes

Register: yes no Make Call Without Reg: yes no

Register Expires: Ans Call Without Reg: yes no

Use DNS SRV: yes no DNS SRV Auto Prefix: no yes

Proxy Fallback Intvl: Proxy Redundancy Method:

Subscriber Information

Display Name: User ID:

Password: Use Auth ID: no yes

Auth ID:

Mini Certificate:

SRTP Private Key:

Audio Configuration

Preferred Codec: Use Pref Codec Only: no yes

G729a Enable: yes no G723 Enable: yes no

G726-16 Enable: yes no G726-24 Enable: yes no

G726-32 Enable: yes no G726-40 Enable: yes no

Release Unused Codec: yes no DTMF Process AVT: yes no

Silence Supp Enable: no yes DTMF Tx Method:

Dial Plan

Dial Plan:

Enable IP Dialing: yes no

General

Line Enable: yes no

Share Line Appearance

Share Ext: private shared Shared User ID:

Subscription Expires:

NAT Settings

NAT Mapping Enable: no yes NAT Keep Alive Enable: no yes
 NAT Keep Alive Msg: NAT Keep Alive Dest:

Network Settings

SIP TOS/DiffServ Value: SIP CoS Value: 4
 RTP TOS/DiffServ Value: RTP CoS Value: 7
 Network Jitter Level: Jitter Buffer Adjustment: down

SIP Settings

SIP Transport: UDP TCP SIP Port:
 SIP 100REL Enable: no yes EXT SIP Port:
 Auth Resync-Reboot: yes no SIP Proxy-Require:
 SIP Remote-Party-ID: no yes Referor Bye Delay:
 Refer-To Target Contact: no yes Referee Bye Delay:
 SIP Debug Option: Refer Target Bye Delay:
 Sticky 183: no yes Auth INVITE: no yes

Call Feature Settings

Blind Attn-Xfer Enable: no yes MOH Server:
 Message Waiting: no yes Auth Page: no yes
 Default Ring: Auth Page Realm:
 Conference Bridge URL: Auth Page Password:
 Mailbox ID: Voice Mail Server:
 State Agent: CFWD Notify Serv: no yes
 CFWD Notifier:

Proxy and Registration

Proxy: Use Outbound Proxy: yes no
 Outbound Proxy: Use OB Proxy In Dialog: no yes
 Register: yes no Make Call Without Reg: yes no
 Register Expires: Ans Call Without Reg: yes no
 Use DNS SRV: no yes DNS SRV Auto Prefix: no yes
 Proxy Fallback Intvl: Proxy Redundancy Method:

Subscriber Information

Display Name: User ID:
 Password: Use Auth ID: no yes
 Auth ID:
 Mini Certificate:
 SRTP Private Key:

Audio Configuration

Preferred Codec: G711a G711e G711d
 G729a Enable: yes no G723 Enable: yes no
 G726-16 Enable: yes no G726-24 Enable: yes no
 G726-32 Enable: yes no G726-40 Enable: yes no
 Release Unused Codec: yes no DTMF Process AVT: yes no
 Silence Supp Enable: no yes DTMF Tx Method:

Dial Plan

Dial Plan:
 Enable IP Dialing: yes no