

Grandstream Device Configuration

STATUS

BASIC SETTINGS

ADVANCED SETTINGS

Admin Password: (purposely not displayed for security protection)
SIP Server: (e.g., sip.mycompany.com, or IP address)
Outbound Proxy: (e.g., proxy.myprovider.com, or IP address, if any)
SIP User ID: (the user part of an SIP address)
Authenticate ID: (can be identical to or different from SIP User ID)
Authenticate Password: (purposely not displayed for security protection)
Name: (optional, e.g., John Doe)

Advanced Options:

Preferred Vocoder: choice 1: choice 2: choice 3: choice 4: choice 5: choice 6: choice 7: choice 8:
(in listed order)
G723 rate: 6.3kbps encoding rate 5.3kbps encoding rate
iLBC frame size: 20ms 30ms
iLBC payload type: (between 96 and 127, default is 97)
Silence Suppression: No Yes
Voice Frames per TX: (up to 10/20/32/64 for G711/G726/G723/other codecs respectively)
Layer 3 QoS: (Diff-Serv or Precedence value)
Layer 2 QoS: 802.1Q/VLAN Tag 802.1p priority value (0-7)
Allow incoming SIP messages from SIP proxy only: No Yes
Use DNS SRV: No Yes
User ID is phone number: No Yes
SIP Registration: Yes No
Unregister On Reboot: Yes No
Register Expiration: (in seconds, default 1 hour, max 45 days)
Early Dial: No Yes (use "Yes" only if proxy supports 484 response)
Allow outgoing call without Registration: No Yes
Dial Plan Prefix: (this prefix string is added to each dialed number)
No Key Entry Timeout: (in seconds, default is 4 seconds)
Use # as Dial Key: No Yes (if set to Yes, "#" will function as the Dial key)
local SIP port: (default 5060)
local RTP port: (1024-65535, default 5004)
Use random port: No Yes
NAT Traversal: No
 Yes, STUN server is: (URI or IP:port)
keep-alive interval: (in seconds, default 20 seconds)
Use NAT IP (if specified, this IP address is used in SIP/SDP message)
Proxy-Require: (if specified, the content will appear in Proxy-Require header)
Voice Mail UserID: (User ID/extension for 3rd party voice mail system)
SUBSCRIBE for MWI: No, do not send SUBSCRIBE for Message Waiting Indication
 Yes, send periodical SUBSCRIBE for Message Waiting Indication
Auto Answer: No Yes
Offhook Auto-Dial: (User ID/extension to dial automatically when offhook)
Enable Call Features: No Yes (if Yes, Call Forwarding & Call-Waiting-Disable are supported locally)
Disable Call-Waiting: No Yes
Send DTMF: in-audio via RTP (RFC2833) via SIP INFO
DTMF Payload Type:
Send Flash Event: No Yes (Flash will be sent as a DTMF event if set to Yes)
Onhook Threshold:
NTP Server: (URI or IP address)
 system ring tone
Default Ring Tone: custom ring tone 1, used if incoming caller ID is
 custom ring tone 2, used if incoming caller ID is
 custom ring tone 3, used if incoming caller ID is
Send Anonymous: No Yes (caller ID will be blocked if set to Yes)
Anonymous Method: Use From Header Use Privacy Header
Time to ring:
Special Feature:
Syslog Server:
Syslog Level:
Firmware Upgrade and Provisioning: Upgrade Via TFTP HTTP
 Firmware Server Path:
 Config Server Path:
 Firmware File Prefix: Firmware File Postfix:
 Config File Prefix: Config File Postfix:
Automatic Upgrade:
 No Yes, check for upgrade every minutes (default 7 days)
 Always Check for New Firmware
 Check New Firmware only when F/W pre/suffix changes
 Always Skip the Firmware Check
Firmware Key: (in Hexadecimal Representation)
Authenticate Conf File: No Yes (cfg file would be authenticated before acceptance if set to Yes)
Lock keypad update: No Yes (configuration update via keypad is disabled if set to Yes)
Allow conf SIP Account in Basic Settings: No Yes
Override MTU Size:

Update

Cancel

Reboot