

Grandstream Device Configuration (Change only Configurations in RED)

STATUS

BASIC SETTINGS

ADVANCED SETTINGS

Admin Password:

(purposely not displayed for security protection)

FXS Port :**SIP Server:****Outbound Proxy:****SIP User ID:**

(Use Account Generated on Smart)

Authenticate ID:

(Use Account Generated on Smart)

Authenticate**Password:**

(Use Password Generated for Account on Smart)

Name:

(optional, e.g., John Doe)

Register Expiration:

(If not set Device will loose registration after 5 Minutes)

local SIP port:**local RTP port:**

(1024-65535, default 5004)

Enable Call Features:

No Yes (Disable Call features)

Send DTMF:

in-audio via RTP (RFC2833) via SIP INFO

DTMF Payload Type:**FXO Port :****SIP Server:****Outbound Proxy:****SIP User ID:**

(Use Account Generated on Smart)

Authenticate ID:

(Use Account Generated on Smart)

Authenticate**Password:**

(Use Password Generated for Account on Smart)

Name:

(optional, e.g., John Doe)

Register Expiration:

(If not set Device will loose registration after 5 Minutes)

local SIP port:**local RTP port:**

(1024-65535, default 5008)

Enable Call Features:

No Yes (Disable Call features)

Send DTMF:

in-audio via RTP (RFC2833) via SIP INFO

DTMF Payload Type:**Number of Rings:**

(Change to 2 Rings)

General Misc. Settings:

<i>Preferred Vocoder:</i> (in listed order)	choice 1:	(Set First Codec)	
	choice 2:	(Set Second Codec)	
	choice 3:		
	choice 4:		
	choice 5:		
	choice 6:	(Set last Codec)	
<i>Voice Frames per TX:</i>	(30 Frames)		
<i>G723 rate:</i>	6.3kbps encoding rate	5.3kbps encoding rate	
<i>iLBC frame size:</i>	20ms	30ms	
<i>iLBC payload type:</i>	(between 96 and 127, default is 97)		
<i>Silence Suppression:</i>	No	Yes	
<i>SIP Registration:</i>	Yes	No	
<i>Unregister On Reboot:</i>	Yes	No	
<i>Layer 3 QoS:</i>	(Diff-Serv or Precedence value)		
<i>Layer 2 QoS:</i>	802.1Q/VLAN Tag	802.1p priority value	(0-7)
<i>Use DNS SRV:</i>	No	Yes	
<i>User ID is phone number:</i>	No	Yes	
<i>Early Dial:</i>	No	Yes (use "Yes" only if proxy supports 484 response)	
<i>Dial Plan Prefix:</i>	(this prefix string is added to each dialed number)		
<i>No Key Entry Timeout:</i>	(in seconds, default is 4 seconds)		
<i>Use # as Dial Key:</i>	No	Yes (if set to Yes, "#" will function as the "(Re-)Dial" key)	
<i>Use random port:</i>	No	Yes	
<i>NAT Traversal:</i>	No		
	Yes, STUN server is:		
<i>keep-alive interval:</i>	(in seconds, default 20 seconds)		
<i>Use NAT IP message)</i>	(if specified, this IP address is used in SIP/SDP		
<i>Proxy-Require: header)</i>	(if specified, the content will appear in Proxy-Require		
<i>TFTP Upgrade Server:</i>	.	.	(for remote software upgrade and configuration)
<i>SUBSCRIBE for MWI:</i>	No, do not send SUBSCRIBE for Message Waiting Indication		
	Yes, send periodical SUBSCRIBE for Message Waiting Indication		
<i>Offhook Auto-Dial:</i>	(User ID/extension to dial automatically when offhook)		
<i>Send Flash Event:</i>	No	Yes (Flash will be sent as a DTMF event if set to Yes)	
<i>FXS Impedance:</i>			
<i>Caller ID Scheme:</i>			
<i>Onhook Voltage:</i>			

