

Grandstream Device Configuration - Microsoft Internet Explorer

Indirizzo: http://192.168.1.101/index.htm

Grandstream Device Configuration

STATUS **BASIC SETTINGS** **ADVANCED SETTINGS**

MAC Address: 00.0B.82.02.3B.8B
WAN IP Address: 192.168.1.101
Product Model: BT100
Software Version: Program-- 1.0.6.6 Bootloader-- 1.0.1.0 HTML-- 1.0.0.49 VOC-- 1.0.1.0
System Up Time: 3 day(s) 13 hour(s) 58 minute(s)
Registered: Yes
PPPoE Link Up: disabled
 NAT: detected NAT type is restricted cone
NAT Mapped IP: 62.123.208.225
NAT Mapped Port: 41060
Total Inbound Calls: 0
Total Outbound Calls: 0
Total Missed Calls: 1
Total Call Time (in minutes): 1389828589
Total SIP Message Sent: 586
Total SIP Message Received: 318
Total RTP Packet Sent: 98379
Total RTP Packet Received: 71586
Total RTP Packet Loss: 2924

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Operazione completata Internet

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Indirizzo: http://192.168.1.101/config2.htm

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End User Password: (purposely not displayed for security protection)
IP Address: dynamically assigned via DHCP (default) or PPPoE (will attempt PPPoE if DHCP fails and following is non-blank)
 PPPoE account ID:
 PPPoE password:
Preferred DNS server: . . .
 statically configured as:
 IP Address: . . .
 Subnet Mask: . . .
 Default Router: . . .
 DNS Server 1: . . .
 DNS Server 2: . . .
Time Zone: GMT+1:00 (Paris, Amsterdam, Berlin, Rome, Vienna, Madrid, Warsaw, Brussels) ▾
Daylight Savings Time: No Yes (if set to Yes, display time will be 1 hour ahead of normal time)
 Year-Month-Day
Date Display Format: Month-Day-Year
 Day-Month-Year

Update Cancel Reboot

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Operazione completata Internet

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:
(in listed order)

choice 1:

choice 2:

choice 3:

choice 4:

choice 5:

choice 6:

choice 7:

choice 8:

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)

Use DNS SRV: No Yes

User ID is phone number: No Yes

SIP Registration: Yes No

Unregister On Reboot: Yes No

<i>Register Expiration:</i>	<input type="text" value="60"/> (in minutes. default 1 hour, max 45 days)
<i>Early Dial:</i>	<input checked="" type="checkbox"/> No <input type="checkbox"/> Yes (use "Yes" only if proxy supports 484 response)
<i>Dial Plan Prefix:</i>	<input type="text"/> (this prefix string is added to each dialed number)
<i>No Key Entry Timeout:</i>	<input type="text" value="4"/> (in seconds, default is 4 seconds)
<i>Use # as Dial Key:</i>	<input type="checkbox"/> No <input checked="" type="checkbox"/> Yes (if set to Yes, "#" will function as the "(Re-)Dial" key)
<i>local SIP port:</i>	<input type="text" value="5060"/> (default 5060)
<i>local RTP port:</i>	<input type="text" value="5004"/> (1024-65535, default 5004)
<i>Use random port:</i>	<input type="checkbox"/> No <input checked="" type="checkbox"/> Yes
<i>NAT Traversal:</i>	<input type="checkbox"/> No <input checked="" type="checkbox"/> Yes, STUN server is: <input type="text" value="stun.voip.eutelia.it:3478"/> (URI or IP:port)
<i>keep-alive interval:</i>	<input type="text" value="20"/> (in seconds, default 20 seconds)
<i>Use NAT IP</i>	<input type="text"/> (if specified, this IP address is used in SIP/SDP message)
<i>Proxy-Require:</i>	<input type="text"/> (if specified, the content will appear in Proxy-Require header)
<i>Firmware Upgrade:</i>	<input checked="" type="checkbox"/> Via TFTP Server <input type="text" value="67"/> <input type="text" value="153"/> <input type="text" value="142"/> <input type="text" value="82"/> <input checked="" type="checkbox"/> Via HTTP Server <input type="text" value="gs-firmw are.gratissip.dk/firmw :"/> Automatic HTTP Upgrade: <input type="checkbox"/> No <input checked="" type="checkbox"/> Yes, check for upgrade every <input type="text" value="1"/> days (default 7 days)
<i>Voice Mail UserID:</i>	<input type="text"/> (User ID/extension for 3rd party voice mail system)
<i>SUBSCRIBE for MWI:</i>	<input checked="" type="checkbox"/> No, do not send SUBSCRIBE for Message Waiting Indication <input type="checkbox"/> Yes, send periodical SUBSCRIBE for Message Waiting Indication
<i>Auto Answer:</i>	<input type="checkbox"/> No <input checked="" type="checkbox"/> Yes
<i>Offhook Auto-Dial:</i>	<input type="text"/> (User ID/extension to dial automatically when offhook)
<i>Enable Call Features:</i>	<input type="checkbox"/> No <input checked="" type="checkbox"/> Yes (if Yes, Call Forwarding & Call-Waiting-Disable are supported locally)
<i>Disable Call-Waiting:</i>	<input type="checkbox"/> No <input checked="" type="checkbox"/> Yes
<i>Send DTMF:</i>	<input type="checkbox"/> in-audio <input checked="" type="checkbox"/> via RTP (RFC2833) <input type="checkbox"/> via SIP INFO
<i>DTMF Payload Type:</i>	<input type="text" value="101"/>
<i>Send Flash Event:</i>	<input type="checkbox"/> No <input checked="" type="checkbox"/> Yes (Flash will be sent as a DTMF event if set to Yes)
<i>NTP Server:</i>	<input type="text" value="129.132.2.21"/> (URI or IP address)
<i>Default Ring Tone:</i>	<input checked="" type="checkbox"/> system ring tone <input type="checkbox"/> custom ring tone 1, used if incoming caller ID is <input type="text"/>

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)

Lock keypad update: No Yes (configuration update via keypad is disabled if set to Yes)

Syslog Server:

Syslog Level: