



## Configuración Grandstream HandyTone 286

**MAC Address:** 00.0B.82.01.E5.9A

**Product Model:** HT286

**Software Version:** Program--1.0.5.11 Bootloader--1.0.0.19 HTML--1.0.0.37 VOC--1.0.0.6 >  
detected NAT type is open Internet

**Admin Password:** - 1 } (password to configure this IP phone)

**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:

Use this DNS server (if specified):  .  .

.

statically configured as:

IP Address:  .  .  .

Subnet Mask:  .  .  .

Default Router:  .  .  .

DNS Server 1:  .  .  .

DNS Server 2:  .  .  .

**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:**  **Password SIP**

**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: <input type="text" value="current setting is 'iLBC'"/>
<b>G723 rate:</b>	<input checked="" type="checkbox"/> 6.3kbps encoding rate <input type="checkbox"/> 5.3kbps encoding rate
<b>iLBC frame size:</b>	<input checked="" type="checkbox"/> 20ms <input type="checkbox"/> 30ms
<b>iLBC payload type:</b>	<input type="text" value="99"/> (between 96 and 127, default is 98)
<b>Silence Suppression:</b>	<input checked="" type="checkbox"/> No <input type="checkbox"/> Yes
<b>Voice Frames per TX:</b>	<input type="text" value="2"/> (up to 10/20/32/64 for G711/G726/G723/other codecs respectively)
<b>Layer 3 QoS:</b>	<input type="text" value="48"/> (Diff-Serv or Precedence value)
<b>Layer 2 QoS:</b>	802.1Q/VLAN Tag <input type="text" value="0"/> 802.1p priority value <input type="text" value="0"/> (0-7)
<b>Use DNS SRV:</b>	<input checked="" type="checkbox"/> No <input type="checkbox"/> Yes
<b>User ID is phone number:</b>	<input checked="" type="checkbox"/> No <input type="checkbox"/> Yes
<b>SIP Registration:</b>	<input checked="" type="checkbox"/> Yes <input type="checkbox"/> No
<b>Unregister On Reboot:</b>	<input checked="" type="checkbox"/> Yes <input type="checkbox"/> No
<b>Register Expiration:</b>	<input type="text" value="60"/> (in minutes, default 1 hour, max 45 days)
<b>Early Dial:</b>	<input checked="" type="checkbox"/> No <input type="checkbox"/> Yes (use "Yes" only if proxy supports 484 response)
<b>Dial Plan Prefix:</b>	<input type="text"/> (this prefix string is added to each dialed number)
<b>No Key Entry Timeout:</b>	<input type="text" value="4"/> (in seconds, default is 4 seconds)
<b>Use # as Dial Key:</b>	<input type="checkbox"/> No <input checked="" type="checkbox"/> Yes (if set to Yes, "#" will function as the "(Re-)Dial" key)
<b>local SIP port:</b>	<input type="text" value="5060"/> (default 5060)
<b>local RTP port:</b>	<input type="text" value="16384"/> (1024-65535, default 5004)
<b>Use random port:</b>	<input checked="" type="checkbox"/> No <input type="checkbox"/> Yes
<b>NAT Traversal:</b>	<input type="checkbox"/> No <input checked="" type="checkbox"/> Yes, STUN server is: <input type="text" value="stun.adiptel.com"/> (URI or IP:port)
<b>keep-alive interval:</b>	<input type="text" value="20"/> (in seconds, default 20 seconds)
<b>Use NAT IP</b>	<input type="text" value="IP Pública Actua"/> (if specified, this IP address is used in SIP/SDP message)
<b>TFTP Server:</b>	<input type="text" value="67"/> . <input type="text" value="153"/> . <input type="text" value="142"/> . <input type="text" value="69"/> (for remote software upgrade and configuration)
<b>SUBSCRIBE for MWI:</b>	<input checked="" type="checkbox"/> No, do not send SUBSCRIBE for Message Waiting Indication <input type="checkbox"/> Yes, send periodical SUBSCRIBE for Message Waiting Indication
<b>Offhook Auto-Dial:</b>	<input type="text"/> (User ID/extension to dial automatically when offhook)
<b>Enable Call Features:</b>	<input type="checkbox"/> No <input checked="" type="checkbox"/> Yes (if Yes, Call Forwarding & Do-Not-Disturb are supported locally)
<b>Disable Call-Waiting:</b>	<input checked="" type="checkbox"/> No <input type="checkbox"/> Yes
<b>Send DTMF:</b>	<input checked="" type="checkbox"/> in-audio <input 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 checked="" type="checkbox"/> No <input type="checkbox"/> Yes (Flash will be sent as a DTMF event if set to Yes)
<i>FXS Impedance:</i>	<input (north="" 600="" america)\""="" ohm="" type="text" value="current setting is \"/>
<i>NTP Server:</i>	<input type="text" value="ntp.adiptel.com"/> (URI or IP address)
<i>Time Zone:</i>	<input (us="" eastern="" gmt-5:00="" new="" time,="" type="text" value="current setting is \" york)\""=""/>
<i>Daylight Savings Time:</i>	<input checked="" type="checkbox"/> No <input type="checkbox"/> Yes (if set to Yes, display time will be 1 hour ahead of normal time)
<i>Send Anonymous:</i>	<input checked="" type="checkbox"/> No <input type="checkbox"/> Yes (caller ID will be blocked if set to Yes)
<i>Lock keypad update:</i>	<input checked="" type="checkbox"/> No <input type="checkbox"/> Yes (configuration update via keypad is disabled if set to Yes)