

Настройки VT-101 на примере созданного аккаунта в нашей системе:

ID	0004989954626510	PWD	EHSPyeQb	Удалить
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ID - номер телефона и PWD пароль.

В адресной строке вашего браузера вводим адрес 192.168.1.100 и нажимаем кнопку «Ввод». Должно появиться окно как на картинке:



В окне Password пишем маленькими буквами admin

Теперь появились настройки вашего телефона VT-101 и заполняем их по примеру показанному на этой картинке и не забываем указывать точные ID и PWD выданные нашей системой в Вашем аккаунте иначе работать не будет:

Grandstream Device Configuration	
ADVANCED SETTINGS	
Admin Password:	<input type="text"/> (purposely not displayed for security protection)
SIP Server:	<input type="text" value="sip.mhcom.eu"/> (e.g., sip.mycompany.com, or IP address)
Outbound Proxy:	<input type="text" value="sip.mhcom.eu"/> (e.g., proxy.myprovider.com, or IP address, if any)
SIP User ID:	<input type="text" value="0004989954626510"/> (the user part of an SIP address)
Authenticate ID:	<input type="text" value="0004989954626510"/> (can be identical to or different from SIP User ID)
Authenticate Password:	<input type="text" value="EHSPyeQb"/> (purposely not displayed for security protection)
Name:	<input type="text" value="MHCOM GmbH"/> (optional, e.g., John Doe)

Advanced Options:

<i>Preferred Vocoder: (in listed order)</i>	choice 1: <input type="text" value="G.729A/B"/>
	choice 2: <input type="text" value="G.729A/B"/>
	choice 3: <input type="text" value="G.729A/B"/>
	choice 4: <input type="text" value="G.729A/B"/>
	choice 5: <input type="text" value="G.729A/B"/>
	choice 6: <input type="text" value="G.729A/B"/>
	choice 7: <input type="text" value="G.729A/B"/>
	choice 8: <input type="text" value="G.729A/B"/>
<i>G723 rate:</i>	<input checked="" type="checkbox"/> 6.3kbps encoding rate <input type="checkbox"/> 5.3kbps encoding rate
<i>iLBC frame size:</i>	<input checked="" type="checkbox"/> 20ms <input type="checkbox"/> 30ms
<i>iLBC payload type:</i>	<input type="text" value="97"/> (between 96 and 127, default is 97)
<i>Silence Suppression:</i>	<input checked="" type="checkbox"/> No <input type="checkbox"/> Yes
<i>Voice Frames per TX:</i>	<input type="text" value="2"/> (up to 10/20/32/64 for G711/G726/G723/other codecs respectively)
<i>Layer 3 QoS:</i>	<input type="text" value="48"/> (Diff-Serv or Precedence value)
<i>Layer 2 QoS:</i>	802.1Q/VLAN Tag <input type="text" value="0"/> 802.1p priority value <input type="text" value="0"/> (0-7)
<i>Use DNS SRV:</i>	<input checked="" type="checkbox"/> No <input type="checkbox"/> Yes
<i>User ID is phone number:</i>	<input checked="" type="checkbox"/> No <input type="checkbox"/> Yes
<i>SIP Registration:</i>	<input checked="" type="checkbox"/> Yes <input type="checkbox"/> No
<i>Unregister On Reboot:</i>	<input type="checkbox"/> Yes <input checked="" type="checkbox"/> No
<i>Register Expiration:</i>	<input type="text" value="10"/> (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 checked="" type="checkbox"/> No <input type="checkbox"/> Yes
<i>NAT Traversal:</i>	<input checked="" type="checkbox"/> No
	<input type="checkbox"/> Yes, STUN server is: <input type="text"/> (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="0"/> <input type="text" value="0"/> <input type="text" value="0"/> <input type="text" value="0"/> <input checked="" type="checkbox"/> Via HTTP Server <input type="text"/> Automatic HTTP Upgrade: <input checked="" type="checkbox"/> No <input type="checkbox"/> Yes, check for upgrade every <input type="text" value="7"/> days (default 7 days)
<i>Voice Mail UserID:</i>	<input type="text" value="50000"/> (User ID/extension for 3rd party voice mail system)
<i>SUBSCRIBE for MWI:</i>	<input type="checkbox"/> No, do not send SUBSCRIBE for Message Waiting Indication <input checked="" type="checkbox"/> Yes, send periodical SUBSCRIBE for Message Waiting Indication
<i>Auto Answer:</i>	<input checked="" type="checkbox"/> No <input 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 checked="" type="checkbox"/> No <input type="checkbox"/> Yes
<i>Send DTMF:</i>	<input checked="" 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 checked="" type="checkbox"/> No <input type="checkbox"/> Yes (Flash will be sent as a DTMF event if set to Yes)
<i>NTP Server:</i>	<input type="text"/> (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"/> <input type="checkbox"/> custom ring tone 2, used if incoming caller ID is <input type="text"/> <input type="checkbox"/> custom ring tone 3, used if incoming caller ID is <input type="text"/>
<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 type="checkbox"/> No <input checked="" type="checkbox"/> Yes (configuration update via keypad is disabled if set to Yes)
<i>Syslog Server:</i>	<input type="text"/>
<i>Syslog Level:</i>	current setting is "NONE" <input type="button" value="v"/>

A rectangular button with a light gray background and a thin black border, containing the word "Update" in a black sans-serif font. The button is centered within a horizontal orange bar that spans the width of the page.

Теперь нажимаем на "Update" и в заключении на "Reboot".

Примерно через 30 секунд система будет готова к работе и Вы можете начинать пользоваться нашими услугами телефонной связи.