

Настройки ATA-486 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 192.168.2.1 и нажимаем кнопку «Ввод». Должно появиться окно как на картинке:



В окне Password пишем маленькими буквами admin

Теперь появились настройки вашего адаптера ATA-486 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 ATA-486	
ADVANCED SETTINGS 1	
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)
Home NPA:	<input type="text"/>
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"/>
<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>Fax Mode:</i>	<input checked="" type="checkbox"/> T.38 (Auto Detect)	<input type="checkbox"/> Pass-Through
<i>Layer 3 QoS:</i>	<input type="text" value="48"/>	(Diff-Serv or Precedence value)
<i>Layer 2 QoS (VoIP):</i>	802.1Q/VLAN Tag <input type="text" value="0"/>	802.1p priority value <input type="text" value="0"/> (0-7)
<i>Layer 2 QoS (PC):</i>	802.1Q/VLAN Tag <input type="text" value="0"/>	802.1p priority value <input type="text" value="0"/> (0-7)
<i>Allow incoming SIP messages from SIP proxy only:</i>	<input checked="" type="checkbox"/> No	<input type="checkbox"/> Yes
<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="600"/>	(in seconds. 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>Allow outgoing call without Registration:</i>	<input checked="" type="checkbox"/> No	<input type="checkbox"/> Yes
<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 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>SIP Registration Failure Retry Wait Time:</i>	<input type="text" value="20"/>	(in seconds. Between 1-3600, default is 20)

NAT Traversal:	<input checked="" type="radio"/> No	<input type="radio"/> Yes, STUN server is: <input type="text"/> (URI or IP:port)
STUN keep-alive interval:	<input type="text" value="20"/> (in seconds, default 20 seconds)	
Use NAT IP	<input type="text"/>	(used in SIP/SDP message if specified)
Use STUN keep-alive to detect networks connectivity:	<input checked="" type="radio"/> No	<input type="radio"/> Yes, total STUN response misses <input type="text" value="5"/> before DHCP restart
Proxy-Require:	<input type="text"/>	
SUBSCRIBE for MWI:	<input checked="" type="radio"/> No, do not send SUBSCRIBE for Message Waiting Indication	<input type="radio"/> Yes, send periodical SUBSCRIBE for Message Waiting Indication
Offhook Auto-Dial:	<input type="text"/>	(User ID/extension to dial automatically when offhook)
Enable Call Features:	<input checked="" type="radio"/> No <input checked="" type="radio"/> Yes	(if yes, call features using star codes will be supported locally)
Use Bell-style 3-way Conference:	<input checked="" type="radio"/> No <input type="radio"/> Yes (if Yes, *23 will be disabled)	
Disable Call-Waiting:	<input checked="" type="radio"/> No <input type="radio"/> Yes	
Disable Call-Waiting Caller-ID:	<input checked="" type="radio"/> No <input type="radio"/> Yes	
Send DTMF:	<input type="checkbox"/> in-audio <input checked="" type="checkbox"/> via RTP (RFC2833) <input type="checkbox"/> via SIP INFO	
DTMF Payload Type:	<input type="text" value="101"/>	
Send Flash Event:	<input checked="" type="radio"/> No <input type="radio"/> Yes (Flash will be sent as a DTMF event if set to Yes)	

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.