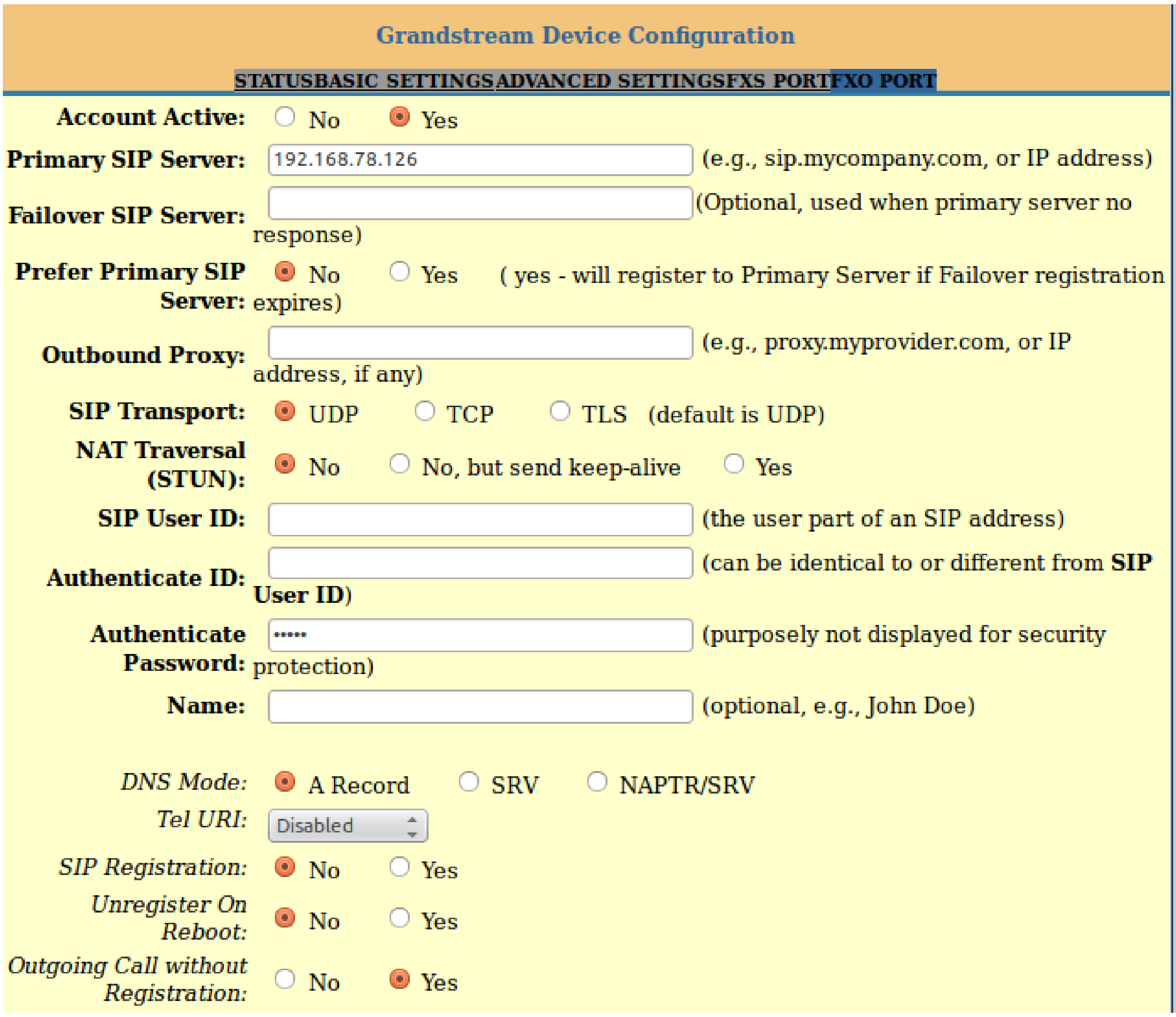
<u>Usando o GXV3500 para receber ligações de um HT-503 com um viva-voz</u>

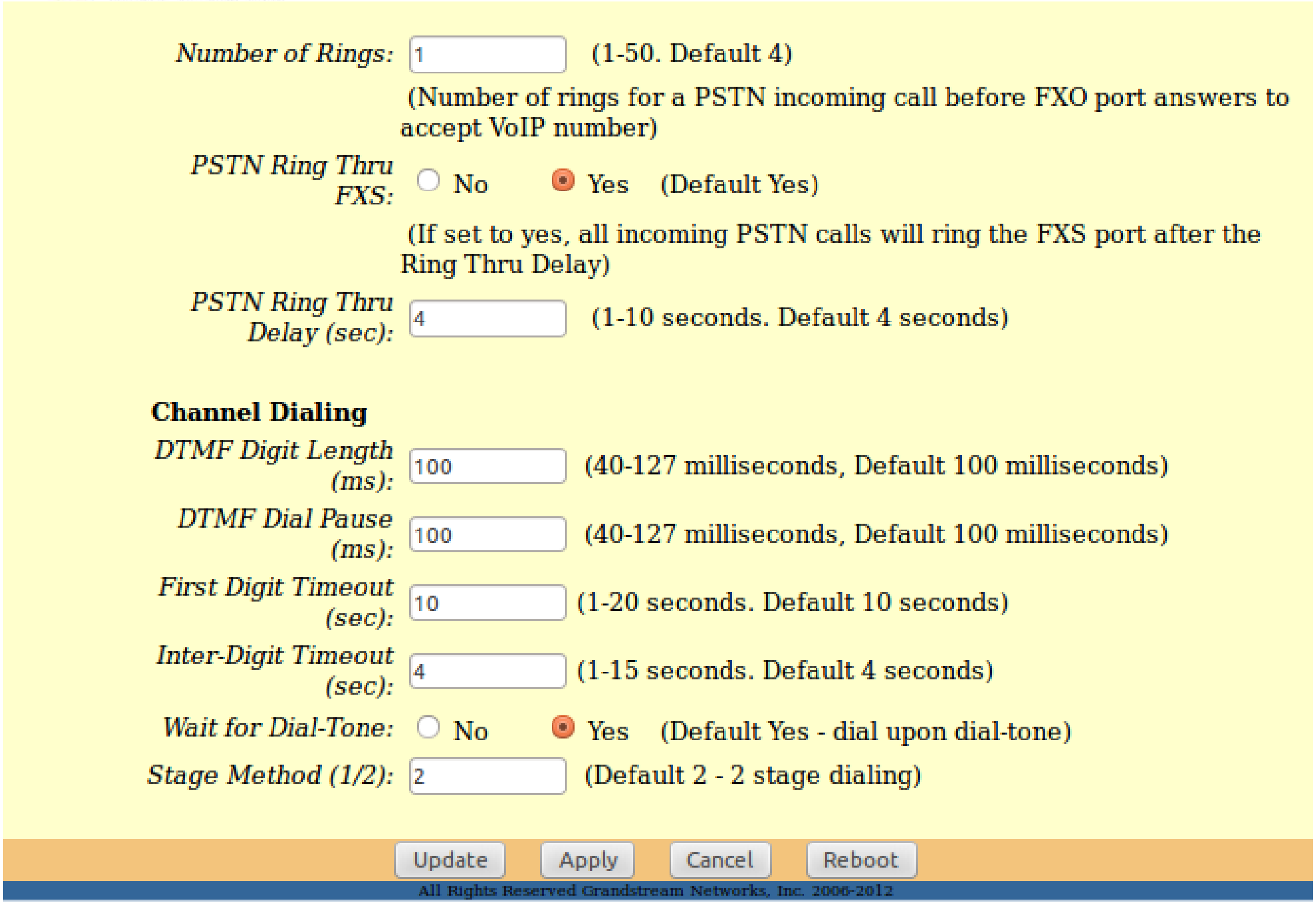
Esse tutorial tem como objetivo utilizar a saída de áudio do GXV3500 para a transmissão da voz proveniente de uma linha/ramal do HT503.

Configuração do HT-503:

Aba **FXO PORT**

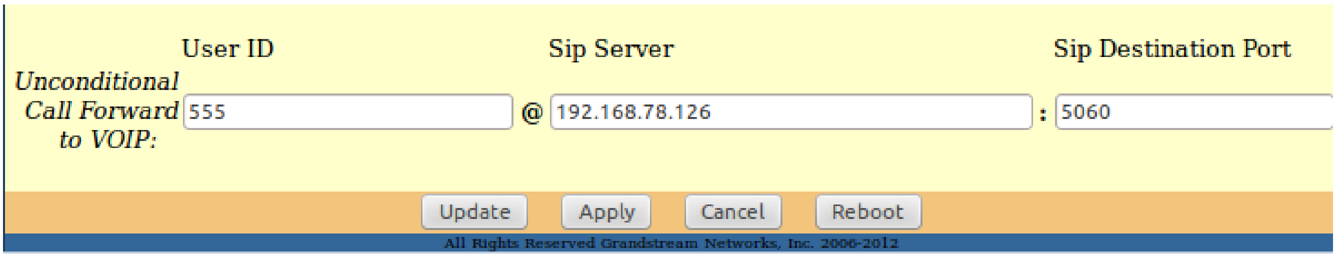
1. Campo ***"Primary SIP server"***: IP do GXV 3500;
2. Campo ***"SIP registration"***: NO
3. Campo ***"Preferred Codec"***: PCMU
4. Campo ***"Number of Rings"***: 1
5. Campo ***"Wait for Dial-Tone"***: NO
6. Campo ***"Stage Method"***: 2


Grandstream Device Configuration
STATUS BASIC SETTINGS ADVANCED SETTINGS FXS PORT FXO PORT
Account Active: No Yes
Primary SIP Server: 192.168.78.126 (e.g., sip.mycompany.com, or IP address)
Failover SIP Server: (Optional, used when primary server no response)
Prefer Primary SIP Server: No Yes (yes - will register to Primary Server if Failover registration expires)
Outbound Proxy: (e.g., proxy.myprovider.com, or IP address, if any)
SIP Transport: UDP TCP TLS (default is UDP)
NAT Traversal (STUN): No No, but send keep-alive Yes
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)
DNS Mode: A Record SRV NAPTR/SRV
Tel URI: Disabled
SIP Registration: No Yes
Unregister On Reboot: No Yes
Outgoing Call without Registration: No Yes



Aba **BASIC SETTINIGS**

1. Campo ***“Uncoditional Call forwarding to VoIP”***: qualquer número
2. Campo ***“Sip Server”***: IP do GXV3500



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Configurações do GXV-3500:

Aba SIP

1. Campo ***“SIP Server”***: IP do HT-503 **<u>com a porta 5062</u>**
2. Campo ***“Preferred Codec”***: PCMU