

## PAPT2-NA

Para acessar as configurações do equipamento, você deverá pegar o endereço IP, discando pelo telefone (plugado no Phone 1) 4 vezes \* e 110#. Ao pegar o endereço IP, digite no internet explorer para acessá-lo.

Ao pegar o endereço IP, digite no internet explorer para acessá-lo, ex:

Endereço IP adquirido na função \*\*\*\* e logo após 110# : 192.168.1.15

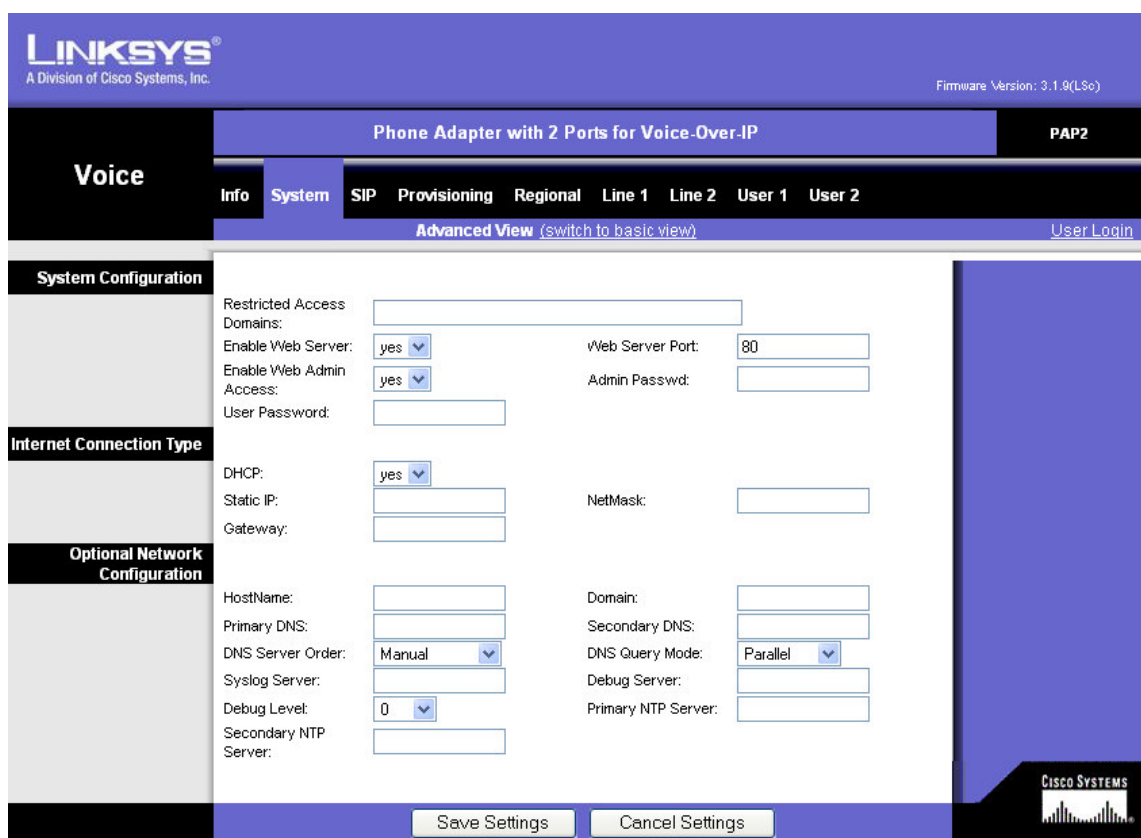
Digite no Internet Explorer o endereço <http://192.168.1.15>, conforme a figura abaixo:



Para Configurar o seu PAP2-NA, recomendamos que você acesse o seu ATA, clique em **Admin Login** (situado no canto superior a direita) e **Switch to Advanced View** (situado abaixo de Line 1) e siga as configurações abaixo:

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- Clique em System e configure os parâmetros de REDE.
- OBS: lembrando que o ATA PAPT2-NA não configura o parâmetro PPPoE.**



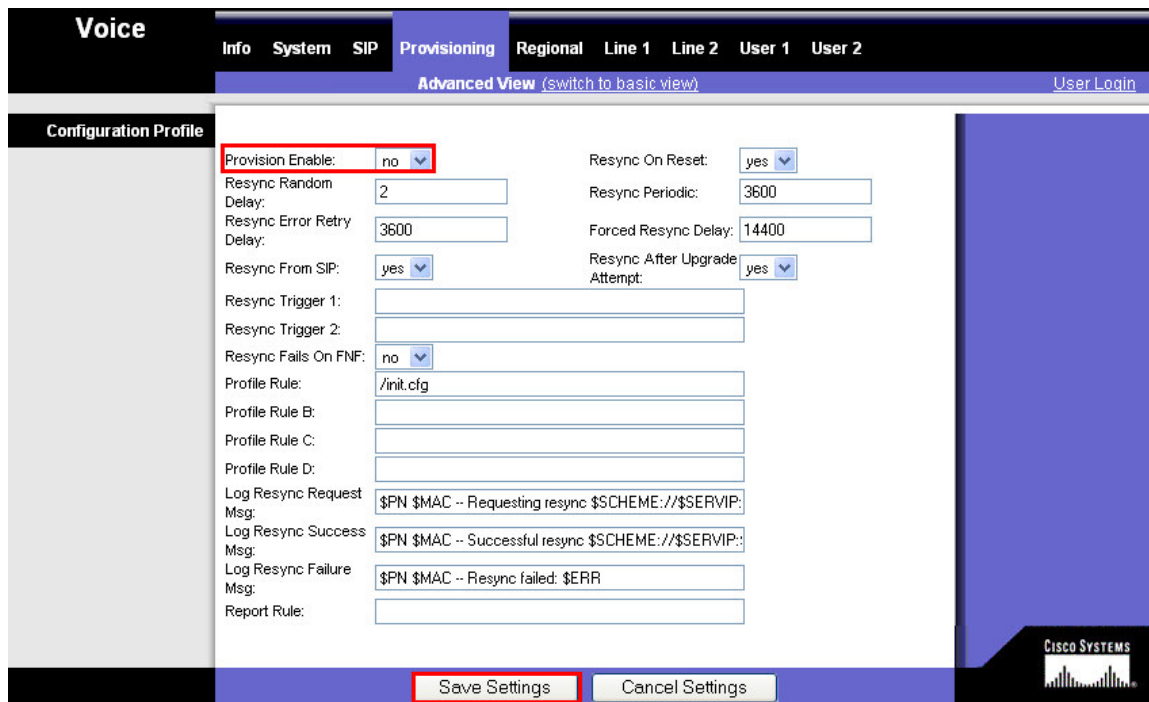
The screenshot displays the configuration interface for a Linksys PAPT2-NA device. The page title is "Phone Adapter with 2 Ports for Voice-Over-IP" and the firmware version is 3.1.9(L5c). The navigation menu includes "Voice", "Info", "System", "SIP", "Provisioning", "Regional", "Line 1", "Line 2", "User 1", and "User 2". The "System" tab is selected, showing "Advanced View" and a "User Login" link.

The configuration page is divided into three main sections:

- System Configuration:** Includes fields for "Restricted Access Domains", "Enable Web Server" (set to yes), "Web Server Port" (80), "Enable Web Admin Access" (set to yes), "Admin Passwd", and "User Password".
- Internet Connection Type:** Includes "DHCP" (set to yes), "Static IP", "Gateway", and "NetMask".
- Optional Network Configuration:** Includes "HostName", "Domain", "Primary DNS", "Secondary DNS", "DNS Server Order" (set to Manual), "DNS Query Mode" (set to Parallel), "Syslog Server", "Debug Server", "Debug Level" (set to 0), and "Primary NTP Server".

At the bottom, there are "Save Settings" and "Cancel Settings" buttons. The Cisco Systems logo is visible in the bottom right corner.

- Clique em **Provisioning**, altere a opção **circulada em vermelho** e logo em seguida clique em **Save Settings**;



**Voice** | Info | System | SIP | **Provisioning** | Regional | Line 1 | Line 2 | User 1 | User 2

Advanced View (switch to basic view) | User Login

**Configuration Profile**

Provision Enable:	no	Resync On Reset:	yes
Resync Random Delay:	2	Resync Periodic:	3600
Resync Error Retry Delay:	3600	Forced Resync Delay:	14400
Resync From SIP:	yes	Resync After Upgrade Attempt:	yes
Resync Trigger 1:			
Resync Trigger 2:			
Resync Fails On FNF:	no		
Profile Rule:	/init.cfg		
Profile Rule B:			
Profile Rule C:			
Profile Rule D:			
Log Resync Request Msg:	\$PN \$MAC -- Requesting resync \$SCHEME://\$SERVIP:		
Log Resync Success Msg:	\$PN \$MAC -- Successful resync \$SCHEME://\$SERVIP:		
Log Resync Failure Msg:	\$PN \$MAC -- Resync failed: \$ERR		
Report Rule:			

Save Settings | Cancel Settings

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- Ao voltar a tela, clique em **Line 1** altere as opções **circuladas em vermelho** e logo em seguida clique em **Save Settings**;

Voice		Info	System	SIP	Provisioning	Regional	Line 1	Line 2	User 1	User 2
							Advanced View (switch to basic view)		<a href="#">User Login</a>	
<b>Streaming Audio Server (SAS)</b>	Line Enable:			yes						
	SAS Enable:			no	SAS DLG Refresh Intvl:			30		
	SAS Inbound RTP Sink:									
<b>NAT Settings</b>	NAT Mapping Enable:			no	NAT Keep Alive Enable:			no		
	NAT Keep Alive Msg:			\$NOTIFY	NAT Keep Alive Dest:			\$PROXY		
<b>Network Settings</b>	SIP TOS/DiffServ Value:			0x68	Network Jitter Level:			high		
	RTP TOS/DiffServ Value:			0xb8	Jitter Buffer Adjustment:			up and down		
<b>SIP Settings</b>	SIP Port:			5060	SIP 100REL Enable:			no		
	EXT SIP Port:				Auth Resync-Reboot:			yes		
	SIP Proxy-Require:				SIP Remote-Party-ID:			no		
	SIP GUID:			no	SIP Debug Option:			none		
	RTP Log Intvl:			0	Restrict Source IP:			no		
	Referor Bye Delay:			4	Refer Target Bye Delay:			0		
	Referee Bye Delay:			0	Refer-To Target Contact:			no		
	Sticky 183:			no						
<b>Call Feature Settings</b>	Blind Attn-Xfer Enable:			no	MOH Server:					
	Xfer When Hangup Conf:			yes	Conference Bridge URL:					
	Conference Bridge Ports:			3						
<b>Proxy and Registration</b>	Proxy:			voipdobrasil.net.br	Use Outbound Proxy:			yes		
	Outbound Proxy:			voipdobrasil.net.br	Use OB Proxy In Dialog:			yes		
	Register:			yes	Make Call Without Reg:			no		
	Register Expires:			3600	Ans Call Without Reg:			no		
	Use DNS SRV:			yes	DNS SRV Auto Prefix:			no		
	Proxy fallback Intvl:			3600	Proxy Redundancy Method:			Normal		
	Voice Mail Server:									
<b>Subscriber Information</b>	Display Name:			Seu Nome	User ID:			Login		
	Password:			Senha	Use Auth ID:			no		
	Auth ID:									
	Mini Certificate:									
	SRTP Private Key:									

Supplementary Service Subscription	
Call Waiting Serv:	yes
Block ANC Serv:	yes
Cfwd All Serv:	yes
Cfwd No Ans Serv:	yes
Cfwd Last Serv:	yes
Accept Last Serv:	yes
CID Serv:	yes
Call Return Serv:	yes
Three Way Call Serv:	yes
Attn Transfer Serv:	yes
MVM Serv:	yes
Speed Dial Serv:	yes
Referral Serv:	yes
Service Announcement Serv:	no
Block CID Serv:	yes
Dist Ring Serv:	yes
Cfwd Busy Serv:	yes
Cfwd Sel Serv:	yes
Block Last Serv:	yes
DND Serv:	yes
CW/CID Serv:	yes
Call Back Serv:	yes
Three Way Conf Serv:	yes
Unattn Transfer Serv:	yes
VMM Serv:	yes
Secure Call Serv:	yes
Feature Dial Serv:	yes

Audio Configuration	
Preferred Codec:	G729a
Use Pref Codec Only:	no
G729a Enable:	yes
G723 Enable:	yes
G726-16 Enable:	yes
G726-24 Enable:	yes
G726-32 Enable:	yes
G726-40 Enable:	yes
FAX Codec Symmetric:	yes
DTMF Tx Method:	Auto
Hook Flash Tx Method:	None
Release Unused Codec:	yes
Silence Supp Enable:	no
Silence Threshold:	medium
Echo Canc Enable:	yes
Echo Canc Adapt Enable:	yes
Echo Supp Enable:	yes
FAX CED Detect Enable:	yes
FAX CNG Detect Enable:	yes
FAX Passthru Codec:	G711u
FAX Passthru Method:	NSE
FAX Process NSE:	yes
FAX Disable ECAN:	no

Dial Plan	
Dial Plan:	[*xx**]x([3469]110 00 2-9)xxxxxx1xxx[2-9]xxxxxS0 xxxx
Enable IP Dialing:	no
Emergency Number:	

FXS Port Polarity Configuration	
Idle Polarity:	Forward
Caller Conn Polarity:	Forward
Callee Conn Polarity:	Forward

Assim o seu ATA estará configurado e funcionando corretamente com a Voip do Brasil. Caso haja alguma irregularidade, recomendamos que você tente refaça a configuração.