

SOLUTION POUR CALLSHOP



CONFIGURATION DE COMPTE SIP

CONFIGURATION LINE1 (Cabine1) :

Pour configurer votre gateway avec un compte SIP, vous devez vous rendre sur l'onglet **Line1** de l'interface d'administration de votre gateway, ensuite procéder comme le montre la figure ci-dessous pour la ligne1 :

The screenshot shows the configuration page for a SIP account on Line 1. The interface is titled "Phone Adapter with 2 Ports for Voice-Over-IP" and "PAP2". The navigation tabs include "Info", "System", "SIP", "Regional", "Line 1", "Line 2", "User 1", and "User 2". The "Line 1" tab is selected, and the "Basic View" is active. The left sidebar contains sections for "SIP Settings", "Proxy and Registration", "Subscriber Information", "Supplementary Service Subscription", and "Audio Configuration".

Configuration du compte SIP pour la la cabine 1 (Line1)

Line Enable: yes

SIP Port: (SIP Port 5060 pour Line 1)

Proxy: (Adresse Proxy)

Make Call Without Reg: yes

Ans Call Without Reg: yes

Display Name:

Password: (Mot de passe de votre compte SIP)

Auth ID:

Register: yes

Register Expires:

User ID: (User ID de votre compte SIP ici)

Use Auth ID: no

Call Waiting Serv: no

Block ANC Serv: no

Cfwd All Serv: no

Cfwd No Ans Serv: no

Cfwd Last Serv: no

Accept Last Serv: no

CID Serv: no

Call Return Serv: no

Three Way Call Serv: no

Attn Transfer Serv: no

MVM Serv: no

Block CID Serv: no

Dist Ring Serv: no

Cfwd Busy Serv: no

Cfwd Sel Serv: no

Block Last Serv: no

DND Serv: no

CWCID Serv: no

Call Back Serv: no

Three Way Conf Serv: no

Unattn Transfer Serv: no

VMM Serv: no

Preferred Codec:

Use Pref Codec Only: yes

DTMF Tx Method:

Silence Supp Enable: no

FAX CED Detect Enable: yes

Buttons: Save Settings, Cancel Settings

Cisco Systems logo

Configuration ligne 2 : (Cabine2)

Pour configurer votre gateway avec un compte SIP, vous devez vous rendre sur l'onglet **Line 2** de l'interface d'administration de votre gateway, ensuite procéder comme le montre la figure ci-dessous pour la ligne2 :

Phone Adapter with 2 Ports for Voice-Over-IP PAP2

Voice

Info System SIP Regional Line 1 **Line 2** User 1 User 2

Basic View (switch to advanced view) User Login

SIP Settings

Line Enable: **SIP Port 5061 pour Line 2**

SIP Port: **Adresse Proxy**

Proxy: **Configuration du compte SIP pour la cabine 2 (Line2)**

Make Call Without Reg: Register:

Ans Call Without Reg: Register Expires:

Subscriber Information

Mot de passe de votre compte SIP

Display Name: **User ID de votre compte SIP ici**

Password: User ID:

Auth ID: Use Auth ID:

Supplementary Service Subscription

| | | | |
|----------------------|---------------------------------|-----------------------|---------------------------------|
| Call Waiting Serv: | <input type="text" value="no"/> | Block CID Serv: | <input type="text" value="no"/> |
| Block ANC Serv: | <input type="text" value="no"/> | Dist Ring Serv: | <input type="text" value="no"/> |
| Cfwd All Serv: | <input type="text" value="no"/> | Cfwd Busy Serv: | <input type="text" value="no"/> |
| Cfwd No Ans Serv: | <input type="text" value="no"/> | Cfwd Sel Serv: | <input type="text" value="no"/> |
| Cfwd Last Serv: | <input type="text" value="no"/> | Block Last Serv: | <input type="text" value="no"/> |
| Accept Last Serv: | <input type="text" value="no"/> | DND Serv: | <input type="text" value="no"/> |
| CID Serv: | <input type="text" value="no"/> | CWCID Serv: | <input type="text" value="no"/> |
| Call Return Serv: | <input type="text" value="no"/> | Call Back Serv: | <input type="text" value="no"/> |
| Three Way Call Serv: | <input type="text" value="no"/> | Three Way Conf Serv: | <input type="text" value="no"/> |
| Attn Transfer Serv: | <input type="text" value="no"/> | Unattn Transfer Serv: | <input type="text" value="no"/> |
| MVM Serv: | <input type="text" value="no"/> | VMM Serv: | <input type="text" value="no"/> |

Audio Configuration

Preferred Codec: Silence Supp Enable:

Use Pref Codec Only: FAX CED Detect Enable:

DTMF Tx Method:

