

Account		Account 1	
Basic >>			
Register Status	Registered		
Account Active	<input checked="" type="radio"/> On <input type="radio"/> Off		
Label	Phone Display Name		?
Display Name			?
Register Name	1xxx8839xxx		?
User Name	1xxx8839xxx		?
Password	*****		?
SIP Server	voip.freephoneline.ca	Port	5060 ?
Enable Outbound Proxy Server	Disabled		?
Outbound Proxy Server		Port	5060 ?
Transport	UDP		?
Backup Outbound Proxy Server		Port	5060 ?
NAT Traversal	STUN		?
STUN Server		Port	?
Voice Mail			?
Proxy Require			?
Anonymous Call	Off		?
On Code			?
Off Code			?
Anonymous Call Rejection	Off		?
On Code			?
Off Code			?
Missed Call Log	Enabled		?
Auto Answer	Disabled		?
Ring Tone	Common		?
Codecs >> ?			
Advanced >>			
Confirm		Cancel	

NOTE

Display Name
SIP service subscriber's name which will be used for Caller ID display.

Register Name
SIP service subscriber's ID used for authentication.

User Name
User account, provided by VoIP service provider.

NAT Traversal
Defines the STUN server will be active or not.

Proxy Require
A special parameter just for Nortel server. If you login to Nortel server, the value should be:
com.nortelnetworks.firewall

Codecs
Choose the codecs you want to use.

Advanced
The Advanced parameters for administrator.

Account Account 1	
Basic >>	
Codecs >> ?	
<div>Disabled Codecs</div> <div>G723_53 G723_63 G722 G726-16 G726-24 G726-32 G726-40</div>	<div>Enabled Codecs</div> <div>PCMU G729 PCMA</div>
Advanced >>	
<div>Confirm</div> <div>Cancel</div>	

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Advanced >>

UDP Keep-alive Message	Enabled	?
UDP Keep-alive Interval (seconds)	20	
Login Expire (seconds)	3600	?
Local SIP Port	5060	?
Rport	Enabled	?
SIP Session Timer (seconds) T1	0.5	?
SIP Session Timer (seconds) T2	4	
SIP Session Timer (seconds) T4	5	
Subscribe Period (seconds)	3600	?
DTMF Type	RFC2833	?
How to INFO DTMF	Disabled	
DTMF Payload	101	
100 Reliable Retransmission	Disabled	?
Enable Precondition	Disabled	?
Subscribe Register	Disabled	?
Subscribe for MWI	Disabled	?
MWI Subscription Period (seconds)	3600	
Subscribe MWI to VM	Disabled	
Use Session Timer	Disabled	?
Session Timer (seconds)	3600	?
Refresher	UAC	?
Use user=phone	Disabled	?
Voice Encryption (SRTP)	<input type="radio"/> On <input checked="" type="radio"/> Off	?
Ptime(ms)	20	?
BLF List URI		?
BLF List Pickup Code		?
BLF List Barge In Code		?
Shared Line	Disabled	?
Dialog-Info Call Pickup	Disabled	?
Direct Call Pickup Code		
Group Call Pickup Code		
BLA Subscription Period (seconds)	300	?
SIP Send MAC	Enabled	?

