**Inbound Proxy: inbound27.vitelity.net**

**Outbound Proxy: outbound.vitelity.net**

**Credentials: lill\_dev/HearSpotBark1**

**DIDs: 425-577-6203  
 425-274-8997**

**Vitelity Communications Setup Wizard**  
  
**Please add the following configuration to your /etc/asterisk/sip.conf**  
  
Add the following at the bottom of your sip.conf  
  
[vitel-inbound]  
type=friend  
dtmfmode=auto  
host=inbound27.vitelity.net  
context=inbound  
allow=all  
insecure=very  
  
[vitel-outbound]  
type=friend  
dtmfmode=auto  
host=outbound.vitelity.net  
context=outbound  
insecure=very  
allow=all  
  
**Please add the following configuration to your /etc/asterisk/extensions.conf**  
  
Add the following to the bottom of your extensions.conf  
  
[outbound]  
exten => \_1NXXNXXXXXX,1,Dial(SIP/$\{EXTEN}@vitel-outbound)  
exten => \_011.,1,Dial(SIP/$\{EXTEN}@vitel-outbound)  
  
; e911 must be enabled. see DIDs > NPANXXNXXX > Action > e911  
exten => \_911,1,Dial(SIP/911@vitel-outbound)  
  
[inbound]  
exten => 4255776203,1,Answer   
exten => 4252748997,1,Answer   
  
**Note:** This is sample configuration and will not entirely configure your PBX system. The answer commands listed for your DIDs will not route them to your phone, it will simply answer the call then disconnect it. You are responsible for your own internal routing.  
  
For more information regarding Asterisk extension logic, please refer to [www.voip-info.org](http://www.voip-info.org/wiki/view/Asterisk+config+extensions.conf) for further assistance.   
  
In order to complete the setup process and configure your Vitelity account for use with your Asterisk PBX Server, please click next.  
  
**NOTICE: Proceeding to the next step will configure ALL of the DID numbers on your account to route to the selected type of equipment and remove any custom DNIS. You may have to reconfigure the routing on existing DID numbers after completing this process.**