

Model Name:	3802/3804/3806	Object:	<input checked="" type="checkbox"/> H.W. <input type="checkbox"/> S.W.
Subject:	Release SIP version firmware of 3802/04/06: 2fxosip.105/ 4fxosip.105/ 6fxosip.105		
Release content			
Bug fixed			
<ol style="list-style-type: none"> 1. Fix the 38xx shutdown issue when it is under Hotline mode. In the call load testing, the previous version will shutdown sometimes. 2. The local SIP listen port is disappeared in p2p mode. 3. Proxy may change the destination number in request line, but 38 will ignore it because it check the destination number is based on the To header. This version revises it. 4. Disable echo cancellation in greeting to avoid the “hang” problem and increase the stability of Disconnect Tone analyzing. 5. Fix the secondary proxy problem. The from and to URL will still be the primary PX address when the proxy switched to 2nd px. 6. Fix the RFC2833 keypad, payload type 101 error in SDP. The payload type is still keep 96 in SDP of invite message although it is changed to 101. 7. Fix some DHCP error of vxworks. 8. Fix the bug that disconnect tone frequency is not stable. But the ring cadence problem still exists. 			
Feature and Function added			
<ol style="list-style-type: none"> 1. Add the info dtmf. Command: sysconf –keypad 2 2. Add blind transfer feature (received). 3. Add routing function with SIP 6500. User can define a special response message to server when it is busy but get an incoming call. Command: sip –busy 438, default is 486 4. Add the CNG disable feature for G711 codec to fix the strange voice when 38 is using Asterisk proxy. Command: sysconf –silence 0 5. “FXO greeting” repeat 3 times and then return to IDLE when gets a PSTN incoming call. 6. Add IVR disable function, using “dial tone” instead. Command: sysconf –ivr 0 			
Configuration item changed			
Note			
<ol style="list-style-type: none"> 1. The routing table is still not base on “longest match” in this version. 2. Recording greeting still has some problems waiting for resolve. 			