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| Model Name: | JIP4001/JIP4002 | Object: | |
| Subject: | Release SIP Firmware of JIP4001/JIP4002 : sipfxso.100 | | |
| Release content | | | |
| Firmware version | | | |
| <p>The version of this firmware:</p> <p>JIP4002 DSP APP: 48304ce1.310 (For g.729 only)</p> <p>JIP4002 DSP APP: 48304ce2.310 (For g.723 only)</p> <p>JIP4001 DSP APP: 48302ce3.140</p> <p>JIP4002 DSP Kernal: 48304ck.310</p> <p>JIP4001 DSP Kernal: 48302ck.140</p> <p>App: JIP4002A - 2sipfxso723.100 (for g.723 only) 2sipfxso729.100 (for g.729 only)</p> <p> JIP4001A - sipfxso.100</p> <p>2M: JIP4002A - 2m2sipfxso723.100 (for g.723 only) 2m2sipfxso729.100 (for g.729 only)</p> <p> JIP4001A - 2msipfxso.100</p> | | | |
| Changed and New Feature | | | |
| <ol style="list-style-type: none"> 1. Networking <ul style="list-style-type: none"> -- Support Static, DHCP and PPPoE mode. 2. Authentication <ul style="list-style-type: none"> -- Support the Proxy authentication for the number and password. 3. Keypad <ul style="list-style-type: none"> -- Support inband and RFC2833 types. 4. Ring time for the FXO detection <ul style="list-style-type: none"> -- User could define the ring time for the calls from PSTN side. 5. No Answer Forward support <ul style="list-style-type: none"> -- Gateway will transfer the calls to the destination if the port didn't answer. 6. Hotline <ul style="list-style-type: none"> -- Gateway will auto dial the number if the port is off hook. 7. EOD (end of dial) <ul style="list-style-type: none"> -- Gateway will transfer the digit "#" if this function is disabled. 8. Routing Table <ul style="list-style-type: none"> -- The calls will transfer to the destination if the number is matched with the route table. 9. Prefix <ul style="list-style-type: none"> -- Insert or drop the digits for the numbers that users dialed. 10.Pbook <ul style="list-style-type: none"> -- Support Peer to Peer mode dialing plan. 11.voice <ul style="list-style-type: none"> -- CNG, VAD support and g723, g711u/a and g729 series codec support. | | | |

12.Record

-- Record the tone of the disconnect, ring back, greeting and asking tone.

13.Configuration

-- WEB, Telnet and Console configuration support.

14.Upgrade

-- TFTP and FTP upgrade support.

15.Dialing Function

-- One or Two stage dialing support.

(This has to be use if the proxy supports it,too.)