Support         company         products         partners         support         contact           Release Notes for3.1.7(GWc)         Sipura Phone Adapter         SPA-3000 1 Port FXS, 1 Port FXO, 1 Ethernet Interface         Copyright (C) 2003-2005 Sipura Technology Inc.	SIPURA technology, inc.						
<pre>SPA-3000 1 Port FXS, 1 Port FXO, 1 Ethernet Interface Copyright (C) 2003-2005 Sigura Technology Inc. * * * * * * * * * IMPORTANT * * * * * * * * * * * * * * * * * * *</pre>	Support	company	products	partners	support	contact	
<pre>SPA-3000 1 Port FXS, 1 Port FXO, 1 Ethernet Interface Copyright (C) 2003-2005 Sigura Technology Inc. * * * * * * * * * IMPORTANT * * * * * * * * * * * * * * * * * * *</pre>	Release Notes for3.1.7(GWc) Sipu	ra Phone Adapt	er				
<pre>************************************</pre>	SPA-3000 1 Port FXS, 1 Port FXO,	1 Ethernet In	terface				
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<ul> <li>since 3.1.5(GW)</li> <li>Messages with only blank lines received at any <sip port=""> may be mistaken as the last received message at any SIP Port</sip></li> <li>If far end device has silence suppression enabled and is using packet size that is not a multiple of 10 ms (such as 5, 15, 25, etc.), user may hear intermittent noise on the SPA when audio burst stops.</li> <li>SPA uses 127 as the payload type for AVT tone if the inbound INVITE (from caller) does not include "a=rtpmap:101 telephone-event" in the SDP</li> <li>Enabling/Disabling Distinctive Ringing, CID and CWCID using *code does not take effect until the unit is reboot'd.</li> <li>Cannot set <ring delay="" indication=""> to 768ms</ring></li> <li>If valid dialed number is shorter than a feature dial pattern, feature dial pattern may be falsely triggered. For example, if dailed number is 0 and feature dial contains *70, then *70 can be matched and 0 cannot be dialed out</li> </ul>	<ul> <li>* Use of Proprietary Information an</li> <li>* This release note document contai</li> <li>* that is to be used only by Sipura</li> <li>* Any unauthorized disclosure, copy</li> <li>* of this information is prohibited</li> <li>* ALL Internet based discussion for</li> </ul>	d Copyright No ns proprietary Technology cu ing, distribut . This restric ums, e.g. DSLr	tice: informat: stomers. ion, or us tion inclu eports.	* ion * * se * udes *			
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since 3.1.5(GW)				=			
parameter. . Forced RTP Port Number to be the next even number in SDP when <rtp min="" port=""></rtp>	<ul> <li>since 3.1.5(GW)</li> <li>Escape only special characters i parameter.</li> </ul>	n the Refer-to	header "H	Replaces"			

and <EXT RTP Port Min> is set to an odd number

- . Change default value of <SIP Remote-Party-ID> to "yes".
- . Added <Warn PSTN Outgoing Call> option. If set to "yes", unit will play <PSTN Warning Tone> when calling out to the PSTN gateway. Default is no.
- . Added <PSTN Warning Tone> to be played when Line 1 connected to the PSTN gateway (gw0) as an indication that user is now using the PSTN to make the call
- . Fixed DTMF caller-id for Netherland phones: Increased polarity reversal post delay to 240 ms before sending 1st dtmf digit. Use '#' as end digit for Denmark only, use 'D' as start digit for ETSI only.
- . Change <Daylight Saving Time Rule> default to new US rule: start=4/-1/7;end=10/1/7;save=1
- . Added <Voice Mail Server> parameter for Line 1. It takes a host name with optional port number (similar to <Proxy> paramter). If port number is not specified, 5060 is used. If parameter is not blank, the SPA will send a one-time SUBSCRIBE to messagey-summary event package to the given host.
- . If a renewal REGISTER is sent to an address different from the last REGISTER, the SPA will not include preemptive credentials until it is challenged by registrar
- . Change <VMWI Ring Splash Len> default to 0 (from 0.5s)
- . Added ring detection debouncing for more robust ring detection