Release Notes for -- Linksys SPA-921/941 5.1.8 921 -- IP Phone 941 -- IP Phone Copyright (C) 2006 by Linksys, a Division of Cisco Systems, Inc. All Rights Reserved. \* Use of Proprietary Information and Copyright Notice: \* This release note document contains proprietary information \* that is to be used only by Sipura Technology, Linksys(R), \* and Cisco Systems, Inc. customers. Any unauthorized \* \* disclosure, copying, distribution, or use of this \* information is prohibited. This restriction includes \* ALL Internet based discussion forums, e.g. DSLreports. /\*\* NOTICE \*\*/ official release \_\_\_\_\_ New Features \_\_\_\_\_\_ ### Since 5.1.8 - Disable 2 --> 4 line license upgrade feature. Make 4 extensions phone as default - On incoming page, do not automatically put the current calls on hold. Instead mix the incoming audio with that of the currently active calls - improve speakerphone - Phone will reply 200 OK to a NOTIFY/ping event inside a dialog - Added Broadsoft shared call appearance bridging support ### version: 5.1.5 - improve speakerphone - Phone will reply 200 OK to a NOTIFY/ping event inside a dialog - Added Broadsoft shared call appearance bridging support ### version: 5.1.4 - Added New parameter in Phone page: <SCA Line Key ID mapping>. - Two options: Vertical First/Horizontal First - Used for Shared Call Appearance Line key ID mapping. Each line key can map two calls. When Vertical First is selected: First SCA line will be mapped to 0, x+0; Second SCA line will be mapped to 1, x+1; ... x means total SCA lines. When Horizontal First is selected: First SCA line will be mapped to 0, 1; Second SCA line will be mapped to 2, 3; ...

### version: 5.1.0

- Added <Auth INVITE> option for Line 1/2 to enable challenging incoming initial INVITE requests

- Three parameters are added: <Reg Retry Random Delay> = Random delay range (in seconds) to add to <Register Retry Intvl> when retrying REGISTER after a failure. Default is 0, which disables this feature

<Reg Retry Long Random Delay> = Random delay range (in seconds) to add to <Register Retry Long Intvl> when retrying REGISTER after a failure. Default is 0, which disables this feature

<Reg Retry Intvl Cap> = The maximum value to cap the expoential backoff retry delay (which starts at <Register Retry Intvl> and doubles every retry. Default value is 0, which disables the exponential backoff feature (i.e., error retry interval is always at <Register Retry Intvl>). If this feature is enabled, <Reg Retry Random Delay> is added on top of the exponential backoff delay value

- SPA should escape occurrences of '#' in the dial string with \$23 in outbound INVITE
- SDP in INVITE response, includes annexb=yes/no if present in INVITE message
- Fixed HTTPS with client authentication out-of-memory condition.
- bug fix: network ring back distortion
- bug fix: Phone does not allow dialing '#' key in the phone number
- If keep alive msg is not a valid \$SIP-METHOD macro and keep alive destination is \$PROXY, and REGISTER is enabled, the keep alive message should be sent to the last Register destionation. But this does not work.
- Fixed this bug: If phone receives a NOTIFY with an unknown event, it might reboot
- Fixed: Phone will not play network ringback even if

early media is negotiated to G711

- Fixed this problem: Phone does not use the same Authorization header fields in the ACK as in the corresponding INVITE, per RFC 3261

### version: 5.1.5

- If keep alive msg is not a valid \$SIP-METHOD macro and keep alive destination is \$PROXY, and REGISTER is enabled, the keep alive message should be sent to the last Register destination. But this does not work.

- Fixed this bug: If phone receives a NOTIFY with an unknown event, it might reboot

### version: 5.1.4

- Fixed this bug: When unit fails over to secondary proxy during INVITE, the correspnding ACK might be sent to the primary if <Proxy Fallback Intvl> expires between the INVITE and the ACK.

- Fixed bug: Dictionary New Entry window softkey wrong display

- Fixed Bug: Call Park Status should not show when Linksys Key System is off.

### version: 5.1.0

- Sending network ringback tone to caller using g729 while there's another call already using g729 will lead to reboot.

- If presumed failed server response after failover, the SPA may take the response from the bad server as if it is from the backup server.

- Include "replaces" in Supported header in outbound SIP messages and accept "replaces" in Require header in inbound SIP messages

- Fixed this bug: NOTIFY request for refer event during call transfer does not include subscription-state header that is required per RFC 3265

- If <NAT Keep Alive Msg> is not \$PING, \$NOTIFY, or \$REGISTER), and <NAT Keep Alive Dest> is \$PROXY, the unit should send the keep alive message to the same address:port as the last REGISTER message. But the unit didn't do that. Instead it resolves the current <Proxy> or <Outbound Proxy> address using DNS A Record and send the keep alive message there.

- Fixed this bug: SPA does not include RTP-Stat header in BYE after the BYE is challenged