



We provide powerful tools for VoIP services and for rapid development of new voice enabled applications

**Technology Features**

- SIP and H.323 signaling support
- Proved interoperability with most VOIP services, IP-PBXs, and Gateways
- Advanced multi-media engine provides superb audio quality
- Multiple service profiles support
- ILS support
- Automatic NAT detection and traversal
- Highly optimized for Intel XScale® and PocketPC
- Multiple DTMF options
- Codecs: G.711 A/u; GSM, Speex, iLBC, G.729
- Operating Systems: Windows 2K/XP, Pocket PC, MAC OS X, Linux
- Easy integration with any USB phones
- Changeable skins
- CallTo and SIP URL support
- Open softphone API

**Administration Features**

- Automated user registration
- Dynamic Provisioning at login and On-Fly Configuration
- Remote management and support
- Announcements
- Account information display
- Localization

**End-user Options**

- Multi-way calling
- Call forward, transfer, hold, and mute
- Call logs, Contact lists, Neighbor locator
- Hot keys and Speed-dial
- Changeable ring-tones and dial-tones

**Fully functional demo is available for download at [www.sjlabs.com](http://www.sjlabs.com).**



<i>Products and services</i>	<i>Description</i>	<i>License</i>
SJphone v.1.50	Light version	Free
SJphone Plus v.1.50	Full version with G.729 codec	Call
SJphone Customized	Basic Customization	Call
Private Label Softphone	Full re-branding with localization	Call
Voice applications development	New features development and integration	Call
Web-dialer	ActiveX web-dialers, download size=400K	Call
Softphone API (beta)	Sample C++ code of the softphone	Call

©1999-2005 SJ Labs, Inc. tel. 1-440-519-1520, fax: 1-440-519-0828, [information@sjlabs.com](mailto:information@sjlabs.com)  
 Specifications are subject to change without notice. Selection, offers, warranties and programs may vary by country.

To schedule a meeting at the VON conference contact us at +1-440-533-5541