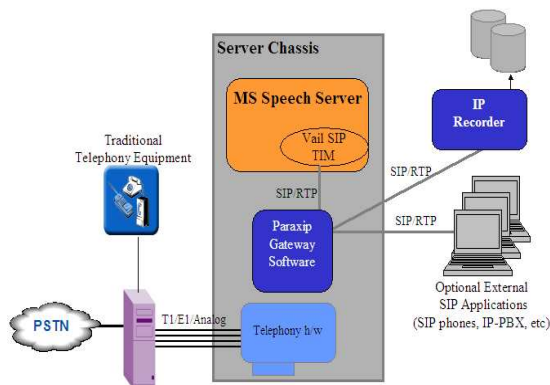


Gateway to Speech

Paraxip offers a portfolio of VoIP gateways that are optimized for speech, IVR and call center applications. Paraxip and partner CYGCOM Integrated Technologies offer a suite of packaged solutions compatible with Microsoft® Speech Server 2004 and the SIP Telephony Interface Manager (TIM) from Vail Systems that enable powerful speech applications in the hybrid network.

Solution Overview

Paraxip Technologies offers software products enabling a seamless transition from traditional telephony to IP telephony. The Paraxip Gateway is a software product that transforms off-the-shelf telephony and media processing boards into speech-optimized media gateways. The gateway is compatible with Microsoft® Speech Server 2004 and the SIP Telephony Interface Manager (TIM) from Vail Systems Inc. The gateway application can run on the same chassis as the speech server or on a different server. The combined solution enables true “hybrid” speech applications that can serve the traditional telephony as well as the IP telephony traffic. The solution further enables significant cost reductions to problems such as call recording, integration to next-generation PBX and more.



Upgrading from Legacy Equipment

Paraxip's software approach enables customers with existing, legacy telephony boards to seamlessly upgrade to Microsoft®

Speech Server 2004 and to a VoIP architecture. Paraxip supports a variety of commercial telephony boards and enables integration via numerous protocols (analog, ISDN, CAS) to a multitude of switches, PBXs and ACDs.

The Paraxip Advantage

Whether the Paraxip Gateway is used to upgrade existing legacy equipment or with packaged hardware and software solutions from CYGCOM, it is the perfect connectivity component to SIP-based Microsoft® Speech Server. The Paraxip Gateway is optimized for the Speech/IVR and Call Center industry with highly differentiated features such as:

- Speech-optimized media processing improving the ROI of speech-recognition applications.
- Sophisticated call transfer capabilities reducing the operational costs associated with self-service applications.
- Advanced Call Progress Analysis (CPA) improving the efficiency and ROI of outbound dialing applications.
- Paraxip Call Forking feature enabling distributed, scalable and reliable call recording and call monitoring.
- Total SIP transparency into the PSTN. Through Paraxip's Routing Engine, SIP applications are no longer trapped behind an opaque media gateway.

For More Information

Please contact Paraxip at info@paraxip.com or visit www.paraxip.com for more details. For sales enquiries, please contact CYGCOM at sales@cygcom.com.