AC4880x Voice over Packet Processor for Client Applications

The **AC4880x** is an ideal low cost voice processing engine for a variety of Voice over IP, Voice over DSL and other voice over packet client applications. The AC4880x VoP processor is a one or two port voice over packet processor that combines toll quality low bit rate voice compression, T.38 compliant fax relay and other voice band processing functions. Field-proven, feature-rich software enables the rapid development and fast time-to-market of the complete solution. The 4880x is based on VoIPerfect™ architecture, AudioCodes' underlying, best-of-breed, core media gateway technology for all of its products.

DELIVER FEATURE-RICH SOLUTIONS

The AC4880x voice over packet processor combines one or two channels of toll quality low bit rate voice compression, with an adaptive G.168-2002 compliant echo canceller, and complete voice band processing functions in a single device. Field-proven G3 Fax Relay, compliant with the T.38 ASN.1 standard, is a major enhancement to the AC4880x offering. Other advanced features include dynamic packet size programming, high quality DTMF/MF R1/MF R2/Call Progress Tones, detection and generation of user defined tones and Caller ID, silence suppression, and automatic voice/fax/data discriminator.

FAST INTEGRATION WITH SOFTWARE STACKS

The AC4880x is supported by the VoicePacketizer™ software stack which enables the processor to create a VoIP-compliant media stream as part of a client entity. The VoicePacketizer™ is an ANSI-C software stack that supports the RTP/RTCP protocol. The software stack also provides a simple API for initialization and configuration of the AC4880x and for run-time call control.

SOFTWARE UPGRADES

The complete functionality of the AC4880x is implemented in the internal DSP software. The software image is downloaded into the AC4880x by the host processor prior to start of operation. This allows the addition of new features in future product releases without any system hardware modifications.

BENEFIT FROM EXTENSIVE EXPERIENCE

AudioCodes is one of the world's leading providers of DSP solutions since 1996. During this period, the company successfully passed numerous interoperability tests while maintaining high levels of performance. AudioCodes' commitment to innovation yields consistently high-quality voice processing products that are feature-rich and field-proven. AudioCodes has deployed over 12 million VoP ports to date.

- Low per-channel cost and power
- Independent channel operation
- Toll quality voice compression
- Robust bandwidth-saving fax relay
- Small footprint
- Proprietary evaluation & development tools
- Lead free available

AC4880x FEATURES
- Direct replacement for AC48301/2
- Low bit rate Vocoders
- Low power consumption & small footprint
- G.168-2002 compliant Echo Canceller
- T.38 compliant Fax Relay
- In Band Signaling detection & generation
- Caller ID detection & generation

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AudioCodes Enabling Technology Products

AC4880x

SPECIFICATIONS

Software Specifications	
Channel Density	One or two low bit rate
Voice Coders	G.729 AB CS-ACELP at 8 kbps
	G.723.1 MP-MLQ at 6.3 kbps, ACELP at 5.3 kbps
	G.727 E-ADPCM at 16-40 kbps
	G.726 ADPCM at 16-40 kbps
	G.711 PCM (µ-Law/A-Law) at 64 kbps
	GSM 6.10 FR at 13 kbps
	NetCoder ® at 6.4-9.6 kbps, 800 bps steps
	Optional: G.728 LD-CELP at 16 kbps
Voice/Fax/Data	Automatic switching
Fax Support	G3 2.4 - 14.4 kbps, T.38 compliant fax relay or automatic switch to PCM
Modem Support	Up to V.92 rates, automatic switch to PCM
Echo Canceller	G.168-2002 compliant
	25 msec tail length
Caller ID Detection	Telcordia (Bellcore) Type 1 & 2,
and Generation	ETSI Type 1 & 2
	NTT Number Display
Three-Way Conferencing	Conferencing of two IP participants with one PSTN participant
In-band Signaling	DTMF (TIA 464B), MFR1,
Detection and Generation	MFR2, AC15, SS4, SS5,
	User Defined and Call Progress tones
Out-of-band Signaling	CAS ABCD (when connected to Standard Framers)
E&M Sampling Resolution	1 msec
Input/Output Gain Control	-31dB +31dB, 1.0 dB steps

Hardware Specifications	
PCM Interface	2.048, 4.096 or 8.192 Mbps,
	µ-Law/A-Law (selectable)
Host Port Interface	8 bit Bidirectional
Power Supply	+1.8V (core), +3.3V (I/O)
Digital Interface Levels	3.3V logic compatible
Power Consumption	180 mW (typ)
Package	144 pin TQFP

APPLICATIONS

- IP-Phone and Key-Phone
- SOHO Voice/data/fax Access Devices
- DSL Access Devices
- Cable Telephony Access Devices
- Voice-enabled Set-top Boxes

ABOUT AUDIOCODES

AudioCodes Ltd. (NASDAQ: AUDC) enables the new voice infrastructure by providing innovative, reliable and cost-effective Voice over Packet technology and Voice Network products to OEMs, network equipment providers and system integrators. AudioCodes provides its customers and partners with a diverse range of flexible, comprehensive media gateway and media processing technologies, based on VoIPerfect™ – AudioCodes' underlying, best-of-breed, core media gateway architecture. The company is a market leader in voice compression technology and is a key originator of the ITU G.723.1 standard for the emerging Voice over IP market. AudioCodes voice network products feature media gateway and media server platforms for packet-based applications in the converged, wireline, wireless, broadband access, and enhanced voice services markets. AudioCodes enabling technology products include VoIP and CTI communication boards, VoIP media gateway processors and modules, and CPE devices. AudioCodes' headquarters and R&D facilities are located in Israel with an R&D extension in the U.S. Other AudioCodes' offices are located in Europe, the Far East, and Latin America.

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