

XVF3800 VOICE PROCESSOR

CRYSTAL CLEAR VOICE CAPTURE FOR COLLABORATION APPLICATIONS

RELIABLE, FAST TIME-TO-MARKET, COST-OPTIMISED SOLUTION THAT ENABLES MICROSOFT TEAMS™ AND ZOOM™ CERTIFICATION FOR ENTERPRISE AND CONSUMER VOICE CONFERENCING PLATFORMS

With remote working on the upward trend, the demand for high-quality, reliable and cost-effective collaboration solutions has increased significantly. Capturing voice clearly and accurately to ensure efficient communication is essential for conferencing platforms to deliver the near face-to-face experience.

Unlike human ears automatically filtering out background noise, microphones capture the whole soundscape including the surrounding and unwanted noise such as acoustic coupling and room reverberation. That's where XMOS cutting-edge algorithms come in.

The XVF3800 4-mic voice processor incorporates state of the art audio algorithms delivering high quality voice in the most challenging of environments. Multi-adaptive beamforming, echo cancellation, de-reverberation and noise suppression ensures outstanding double talk performance and an optimised voice signal ideal for collaboration applications requiring Teams and Zoom certifications.

The addition of XVF3800 will provide your customers with a cost effective high-clarity voice solution with minimal integration effort. XVF3800 delivers a fast time-to-market conferencing solution ideal for enterprise and consumer conferencing platforms.



FEATURE HIGHLIGHTS

The XVF3800 voice processor includes microphone interfacing, voice processing and control, allowing you to fully optimise voice performance based on the specific product acoustics.

ACOUSTIC ECHO CANCELLATION (AEC)

Acoustic Echo Cancellation removes echo from the microphone audio input while reducing far-end echoes enabling XVF3800 to extract clear voice signals even with high-volume speaker output.

NOISE SUPPRESSION

Noise suppression nulls stationary and non-stationary diffuse noise sources, for example air-conditioning and road noise where the frequency characteristics don't change over time. This enables accurate, consistent voice capture.

ADAPTIVE MULTIPLE BEAMFORMERS

The free-running and focused beamformers track both stationary and non-stationary voice sources, identifying the direction of arrival (DoA) and isolating the voice of interest to enhance double talk and dereverberation performance.

AUTOMATIC GAIN CONTROL (AGC)

The configurable and flexible Automatic Gain Control tunes the output channels for optimal voice communication. Ideal for passing Teams and Zoom certifications.

DEV KIT



XK-VOICE-SQ66

APPLICATIONS



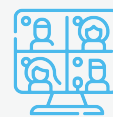
ENTERPRISE
CONFERENCING



VIDEO
BARS

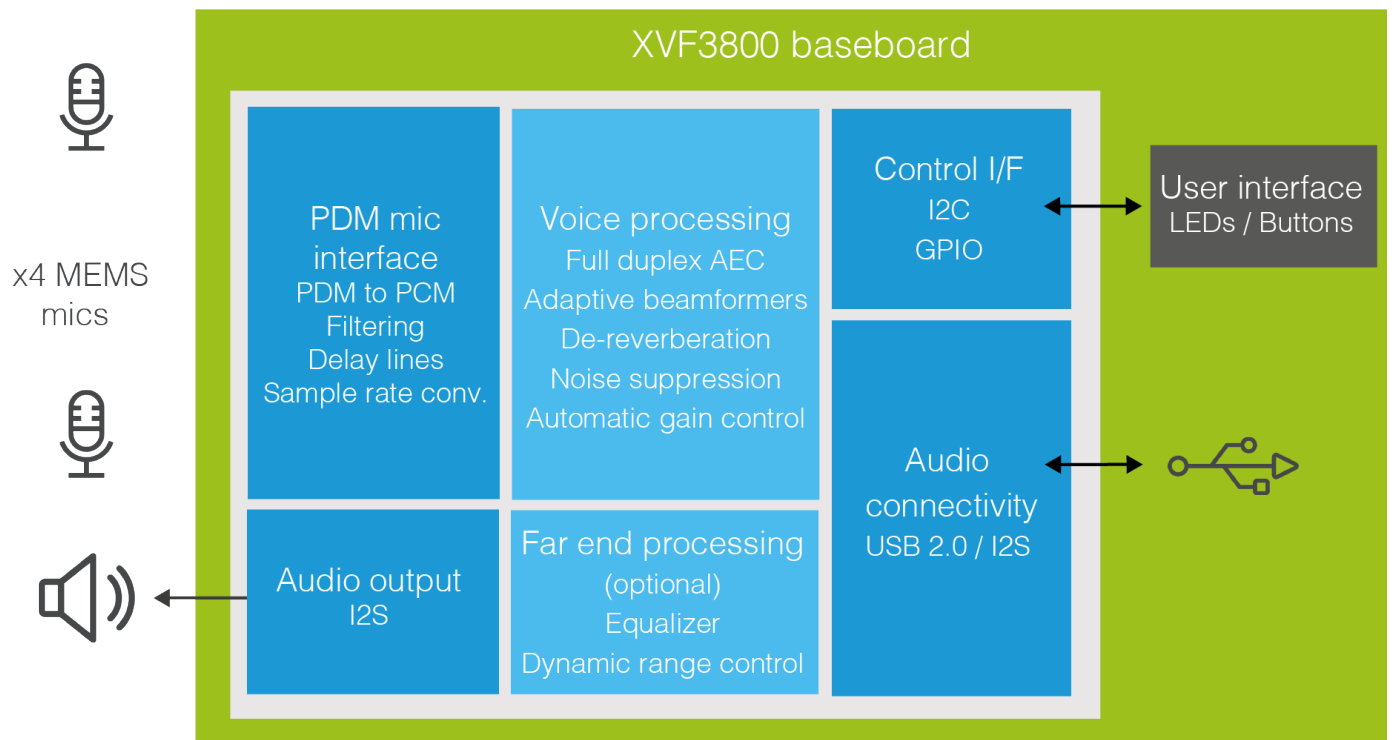


PORTABLE
SPEAKERPHONES



COLLABORATIVE
DISPLAY

BLOCK DIAGRAM



VOICE PROCESSOR | XVF3800

PACKAGE

60-Pin QFN, 0.4mm pitch

VOICE PROCESSING

Acoustic Echo Cancellation
Adaptive Beamformers
Noise Suppression
Automatic Gain Control

MICROPHONE INTERFACE

4 x digital PDM microphone interface
100mm linear mic array, 33mm inter-mic spacing
90mm square mic array, 43mm inter-mic spacing

HOST INTERFACE OPTIONS

High speed USB2.0 device supporting USB Audio
Class 2.0; 16kHz or 48kHz sample rate
I2S audio interface; 16kHz or 48kHz sample rate

AUDIO OUTPUT OPTIONS

I2S output to DAC; 16kHz or 48kHz PCM

CONTROL INTERFACE

USB Control Interface
I2C Control Interface

VOICE PROCESSOR

XVF3800

For further information please contact sales@xmos.com

DEV KIT

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