XVF3610 VOICE PROCESSOR

HIGH PERFORMANCE, LOW BOM

"VOICE IS THE NEXT TECH DISRUPTION. NOTHING IS AS FAST OR NATURAL."

DAVE ISBITSKI, CHIEF ALEXA EVANGELIST, AMAZON

From doorbells to smart TVs, our devices are getting smarter. Virtual Assistants and voice-control have opened the door to a natural conversation with technology; voice is transforming the way in which we access and enjoy content.

To address the demands of this rapid far-field voice market growth, manufacturers need an eBOM-efficient, high-performance voice processor. Requiring just 2 microphones, our XVF3610 voice processor enables far-field voice capture with close range precision, delivering powerful performance in a cost efficient package.

With purpose-designed algorithms, XVF3610 will capture a clear voice stream from across the room, even in noisy environments and when content is streaming through the device.

The XVF3610 is for designers of voice-enabled smart products, who need high performance at a low BOM.



FEATURE HIGHLIGHTS

The XVF3610 offers two firmware variants: XVF3610-UA supports USB accessory devices and XVF3610-INT is designed to enable built-in solutions. Both contain our purpose-designed algorithms.

ACOUSTIC ECHO CANCELLATION (AEC)

Stereo acoustic echo cancellation enables the XVF3610 to detect voice signals even when high-volume audio is playing through the product, enabling barge-in across content. The echo canceller constantly adapts to the room, modeling changes such as people moving, to remove the echoes from the speakers from the microphone input.

INTERFERENCE CANCELLER (IC)

The interference canceller works intelligently to scan the soundscape of the room. It removes static point noise (e.g. from household appliances) and 'ignores' any audio playing from another device. This enables the XVF3610 to capture a clear voice command across a noisy acoustic environment.

NOISE SUPPRESSION

Noise Suppression removes stationary and non-stationary diffuse noise sources, for example air-conditioning and road noise, from the received signal. This enables accurate, consistent voice detection.

AUTOMATIC DELAY ESTIMATION CONTROL (ADEC)

Automatic Delay Estimation Control dynamically monitors reference signal latency and adjusts this to maintain optimal AEC performance in situations where the audio output delay is unknown, such as TVs and STBs.

AUTOMATIC GAIN CONTROL (AGC)

Automatic Gain Control tunes the output channel level for optimum results, whether that's for an Automatic Speech Recognition Service (ASR) or communications applications.

SYSTEM CONTROL AND PARAMETERISATION

Parameterisation of XVF3610 algorithm control, system configuration and GPIO in real-time from host interface, or read from flash memory for default start-up behaviour.

XVF3610-INT / BUILT-IN

XVF3610-UA



APPLICATIONS

TV / SET-TOP BOX ACCESSORY



AUDIO VISUAL PRODUCTS

:::

SMART HOME APPLIANCES



HEALTH AND FITNESS



CONFERENCE CALLING



DEV KIT BLOCK DIAGRAM XVF3610-UA USB ACCESSORY USB Control Interface with high speed USB2.0 compliant audio interface supporting USB Audio Class 1.0 at 48kHz UAC 1.0 sample rate. PDM mic interface XVF3610-INT to apps processor BUILT-IN I2C Serial Control Interface with I2S Audio XVF3610 interface at 48kHz sample rate. 12C Stereo DAC / amp



Shown assembled with RPi (not supplied) for full AST system demonstration.

VOICE PROCESSOR

60-Pin QFN 7mm x 7mm, **PACKAGE**

> 0.4mm pitch QF60A - 1.8V IO

QF60B - 3.3V IO

VOICE **PROCESSING** Full duplex stereo acoustic echo cancellation (225ms tail length)

Automatic Delay Estimator (+/- 150ms delay adjustment)

Interference Canceller Noise Suppression Automatic Gain Control

MICROPHONE INTERFACE

2-channel digital PDM microphone interface

Dual microphone array,

71mm spacing

AUDIO

INTERFACE

High speed USB2.0 compliant device supports

USB Audio Class 1.0 at 16/48kHZ sample rate

I2S audio interface, 16/48kHz

GPIO

4 x General Purpose Inputs, (XVF3610-UA has optional single pin interrupt capability

via USB-HID)

4 x General Purpose Outputs

CONTROL **INTERFACE** **USB** Control Interface

12C Control Interface

TYPICAL POWER CONSUMPTION

USB: 300mW

12S: 300mW

VOICE PROCESSOR

XVF3610-QF60A-C (1v8) XVF3610-QF60B-C (3v3)

USB ACCESSORY

DEV KIT: XK-VOICE-L71 FIRMWARE: XVF3610-UA

BUILT-IN

DEV KIT: XK-VOICE-L71 FIRMWARE: XVF3610-INT

The XVF3610 replaces the XVF3510, which should no longer be used for new designs.

xmos.ai/vocalfusion-voice-interfaces/

