

# **PersonaSound™**

#### HEARING, VOICE, AUDIO, ENHANCEMENT TECHNOLOGIES

for TWS earbuds, headsets, OTC hearing aids & beyond









Revision 2.0 Feb 2024

# Alango DSP SOFTWARE THAT MAKES SOUND BETTER





Founded 2002, HQ in Israel, Haifa Offices in China, Korea, Armenia

35 full time employees

Founder & CEO – Dr. Alexander Goldin

## OVER 70 MILLION PRODUCT LICENSES DELIVERED

BY 2023 AND GROWING FAST...



Automotive hands-free calls & audio



TWS earbuds, headsets & hearables



Audio/video conferencing



Hearing aids and assistive listening



SMART: speakers, TV, appliances

## **PersonaSound™ CONCEPT - ONE SOUND DOES NOT FIT ALL**

WE... have individual hearing profiles • are in distinct places • use different devices





















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## PersonaSound™ MAKES SOUND PERSONAL FOR...

- YOUR HEARING
- YOUR ENVIRONMENT
- YOUR DEVICE



Face-to-face communications



Live music



Music and audio



Remote communications

#### AUGMENTED LISTENING



Sounds of nature



Hearing protection

#### **AUGMENTED HEARING**

## **PersonaSound FOR HEARABLES**



**HEADPHONES** 



TRUE WIRELESS STEREO EARBUDS



**HEARING AIDS** 

## PersonaSound IN ALANGO DSP SOFTWARE PACKAGES



HEARING ENHANCEMENT PACKAGE (HEP) Adaptive Beamforming Noise Reduction Feedback cancellation Personalized Amplification Own-Voice Attenuation



#### HEARING CONTROL PACKAGE (HCP)

Hearing test Loudness/tone control HEP parameters calculator



Personalized Audio Enhancement Ambient Noise Dependent Equalizer Smart Transparency Sound Effects Normalization



VOICE COMMUNICATION PACKAGE (VCP)

Incoming Voice PersonalizationIncoming Noise ReductionAmbient Noise Dependent Equalizer

# HEARING ENHANCEMENT PACKAGE (HEP)



#### Binaural synchronization improving perceptual quality and sound localization for:

- Hearing Personalization by Wide-band Dynamic Range Compression
- Directivity for conversation focus using 2 synchronized beamforming mic-pairs
- Coordinated left-right ear noise reduction
- Coordinated acoustic feedback cancellation
- Reliable user's own voice detection and attenuation using bone conducted voice vibration sensor

# AUDIO ENHANCEMENT PACKAGE (AEP)



- Audio-Personalization by Wide-band Dynamic Range Compression.
- SonicVibrance<sup>™</sup> technology for stereo expansion, enhancement of spectral components and bass boost.
- VoiceFirst™ technology for reduction of background sound effects and speech intelligibility boost for streamed AV content.
- Automatic Volume and Equalization for adjusting sound according to ambient noise level and spectrum.
- Smart transparency mixer technology for ambient awareness

# VOICE COMMUNICATION PACKAGE (VCP)



- Incoming voice personalization by Wide-band Dynamic Range Compression
- Automatic Volume and frequency Equalization according to ambient noise amplitude and spectrum
- EasyListen™ technology slowing down speech of fast talkers in real time during a phone call
- Dual microphone array for exceptional voice transmit clarity even in noisy environments
- OnlyVoice<sup>™</sup> technology for ultimate noise reduction combining microphone array output with a bone conducted voice vibration sensor signal

## **ALANGO AI SPEECH ENHANCEMENT IN NOISE**

#### ALANGO APPROACH - THE BEST OF BOTH WORLDS

- Slow convergence
- Stationary noises only
- Limited performance in low SNR
- Low computational resources
- Low processing delay
- Easy integration with other technologies

- Fast convergence
- Attenuation of nonstationary noises
- Good performance with low SNR

- High computational resources (MIPS, RAM)
- Processing delay
- Difficulty of integration with other technologies

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## ALANGO ZERO LATENCY SIGNAL PROCESSING (ZLSP) COMING SOON, PATENT PENDING



Time critical, ultra low latency classical DSP signal flow path

## **ALANGO AI TECHNOLOGY DEMOS**

Click on the picture or the link...

#### **NOISE REDUCTION**



https://www.youtube.com/shorts/SEwJXqxF4Ms

#### DEREVERBERATION



https://www.youtube.com/shorts/sSktLnZ8eIA

## HEARING SELF-TEST: ASSESSMENT





Alango to provide supporting examples.



#### START THE HEARING TEST

User hears wobbling test tones of different frequencies, one tone one ear at a time.



FOR EACH TEST SOUND:

The user adjusts the volume until finding his/her hearing threshold (the level at which the tone is barely audible).



AUDIOGRAM & HEARING PROFILE

Upon completion, the thresholds are presented as an audiogram and saved as a hearing profile.



UPLOAD AND APPLY

The hearing profile is uploaded to the device and the user enjoys personalized sound in all device modes (hearing, music, phone calls).

## PersonaSound IS MUCH DIFFERENT THAN FREQUENCY EQUALIZATION

#### **TYPICAL HEARING LOSS**

- Different for different frequencies.
  High frequency loss being the most common.
- Not a simple, equal attenuation of sounds.
  Quiet sounds are attenuated more than loud sounds.

#### EQ IS A WRONG TOOL FOR HEARING LOSS COMPENSATION

• Additional amplification of loud sounds can become overbearing.

#### PersonaSound HEARING LOSS COMPENSATION

- Based on Wide-band Dynamic Range Compression (WDRC) principals.
- Sounds are amplified depending on their frequency and level according to the individual hearing profile.



Parameters of Wide-band Dynamic Range Compressor

## **ALANGO BeHear® PRODUCTS – PERFECTING THE CONCEPT**

#### MAKING SUCCESSFUL PRODUCTS REQUIRES A GOOD TECHNOLOGY PERFECTING A TECHNOLOGY REQUIRES MULTIPLE PRODUCTS

- Self-fit, advanced, personalized listening & hearing amplification
- Full Bluetooth voice/audio functionality optimized for individual hearing
- Assistive listening capabilities with personal sound





**BeHear SMARTO** 



**BeHear ACCESS** 

**BeHear PROXY** 



**BeHear ACTIVE** (coming soon)

## ALANGO BeHear – BETTER THAN ADVANCED HEARING AIDS

Speech Intelligibility Index (SII) demonstrates which percentage of speech is intelligible to the user. More information on SSI: <u>https://blog.ansi.org/2016/06/speech-intelligibility-index</u>

BeHear ACCESS SSI is higher compared to Advanced HA in all three tests at 55dB, 65dB, 75dB.\*



#### **BeHear ACCESS**

LEGEND, RIGHT			
1	SII: [46%]	NAL NL2 – 55 (84) dB – ISTS Si	
2	SII: [63%]	NAL NL2 – 65 (90) dB – ISTS Si	
3	SII: [71%]	NAL NL2 – 75 (93) dB – ISTS Si	

Advanced Hearing Aids (a top model from one of big 5)

LEGEND, RIGHT			
1	SII: [42%]	NAL NL2 – 55 (82) dB – ISTS Si	
2	SII: [54%]	NAL NL2 – 65 (86) dB – ISTS Si	
3	SII: [63%]	NAL NL2 – 75 (89) dB – ISTS Si	

\* Full report available upon request

## **ALANGO UNIQUE TECHNOLOGIES**

- Ultra low complexity Neural Networks:
  - Noise reduction
  - Dereverberation
- Patent pending: Zero Latency Signal Processing (ZLSP) combining high frequency resolution with ultra-low delay signal path
- Patent pending: Own Voice Attenuation (OVA) using a bone conduction voice sensor
- EasyListen™ slowing down speech in real time
- Blind Interference Cancellation (BIC) with dual or multimicrophone arrays



## WHAT MAKES ALANGO NUMBER ONE

Alango Technologies is the only company having:

- Comprehensive portfolio of proven technologies (hearing, voice, audio)
- Established relevant customer base of major OEMs
- Established partnership networks with major chipset providers and ODM manufacturers
- Line of BeHear concept products that users love
  <u>https://alango-behear.com</u>
- 20 years of successful experience in DSP software licensing



## PersonaSound BEYOND HEARABLES

We can help making all these devices to sound personal:









TV, SOUNDBAR

CAR

PC

**STEREO** 

## **THANK YOU!**



General enquires - info@alango.com

Sales enquiries - <u>sales@alango.com</u>

Engineering enquiries - tech@alango.com

#### DON'T HESITATE TO CONTACT US. WE ARE ALWAYS LOOKING FORWARD TO HEARING FROM YOU!

Email us your questions, comments, thoughts, or proposals.



Main office: +972 (4) 8580 743 Fax: +972 (4) 8580 621



www.alango.com

**APPENDIX** 

## **APPENDIX**

#### **VOICE TECHNOLOGIES**

<u>EasyListen™</u>

<u>OnlyVoice</u>™

Dynamic Noise Suppression

Wind Noise Reduction

Acoustic Echo Cancellation

Acoustic Feedback Reduction

Adaptive Dual Microphone

Multi-Microphone Arrays

Automatic Volume and eQualization Control

Automatic Gain Control

Multiband Dynamic Range Processing

Voice Activity Detection

AUDIO ENHANCEMENT TECHNOLOGIES

<u>ListenThrough</u>™

<u>SonicVibrance™</u>

Automatic Volume and eQualization Control

Loudspeaker Frequency Response Correction

Multiband Dynamic Range Processing

<u>VoiceFirst</u>™

### EeasyListen™: REAL-TIME SPEECH TEMPO MODIFICATION

Real-time audio processing technology that slows incoming speech, making it more intelligible.

Works by dynamically stretching (slowing) certain parts of speech while adding no noticeable overall delay.

## OnlyVoice™: ULTIMATE VOICE QUALITY FROM EVERYWHERE

Enables robust voice acquisition in True Wireless earbuds from extreme noise conditions.

Works by intelligently mixing the output of adaptive dual microphone beamforming with bone conduction voice sensor.

Can work with in-ear feedback microphone instead of the sensor.

### **DYNAMIC NOISE SUPPRESSION**

Reduces stationary and dynamic noises (e.g. approaching car) in single-channel speech signals without noticeable artefacts.

Dynamic Noise Suppression increases the signal-to-noise ratio and improves speech intelligibility thus reducing listening fatigue.

## WIND NOISE REDUCTION (WNR)

Automatically detects and attenuates wind noise while preserving clear speech.

WNR uses sub-band processing to filter out wind noise in affected frequency regions leaving "good" frequency regions untouched.

## **ACOUSTIC ECHO CANCELLATION (AEC)**

Eliminates acoustic coupling between speaker and microphone, allowing high-quality, full-duplex voice communication without echo.

Alango AEC technology supports mono and stereo echo cancellation with very fast adaptation times.

Alango AEC includes highly efficient, sub-band Acoustic Echo Suppressor.

### **ACOUSTIC FEEDBACK REDUCTION**

Alleviates acoustic feedback by using sub-band adaptive filtering, noise and howling suppression, and quasi-proportional frequency shifting.

### **ADAPTIVE DUAL MICROPHONE**

Creates a directional or noise-canceling microphone using just two omnidirectional microphones.

Adaptively changes mic directionality to achieve best possible noise reduction.

Includes adaptive wind noise suppression.

Allows optional microphone calibration during the production.

### **MULTI-MICROPHONE ARRAYS**

Multi-microphone adaptive beamforming with 2 or more microphones.

Allows multi-beam configurations and sound of arrival detection.

Robust to wind noises.

### **AUTOMATIC VOLUME AND EQUALIZATION CONTROL**

Improves perceptual loudness and intelligibility of voice in dynamic environment with changing noises.

Works by automatically amplifying and equalizing sounds in response to ambient noise characteristics.

### **AUTOMATIC GAIN CONTROL**

Used in conjunction with Dynamic Range Processing to amplify a weak voice signal.

To reduce noise pickup, gain is applied only when user voice activity is detected.

### **VOICE ACTIVITY DETECTION**

Differentiates between voice and ambient noises in an acoustic signal, enabling low-power operation of always-on devices.

The VAD block acts as a low-power "watchdog," alerting the system to use heavier processing blocks only when the user's voice is detected.

### ListenThrough™: ENVIRONMENTAL AUDIO AWARENESS

Improves situational awareness for stereo headset users.

Important external sounds are emphasized to avoid masking by music, while ambient noise is filtered out.

### SonicVibrance™: MUSIC ENHANCEMENT

An integrated set of DSP technologies to enhance the audio experience in music and entertainment applications, including:

- stereo expansion (normalization)
- spectral enhancement,
- bass correction and amplification

### **AUTOMATIC VOLUME AND EQUALIZATION CONTROL**

Increases the perceived loudness and clarity of music being played in noisy environments.

Works by adaptively amplifying and equalizing the music in response to changing ambient noise characteristics.

### LOUDSPEAKER FREQUENCY RESPONSE CORRECTION

Corrects loudspeaker amplitude and phase responses for reproducing natural sound.

### **MULTIBAND DYNAMIC RANGE PROCESSING**

Adjusts the dynamic range of an audio signal by

- splitting into frequency sub-bands,
- processing each band according to specified characteristics, and
- synthesizing the resultant signal.

### **VoiceFirst**<sup>™</sup>

Improves listening experience by enhancing speech intelligibility of TV, movies, and sports broadcasts.

Works by detecting and attenuating background sounds while preserving speech and dialogue.