

# Designing to Meet New Expectations for Audio Experiences and Standards Critical for Competitiveness

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# Agenda

- Introduction
- Far Field Voice and Speech
- Output and Speakers
- Innovation, SoundWire and New Audio Hardware
- Wrap Up

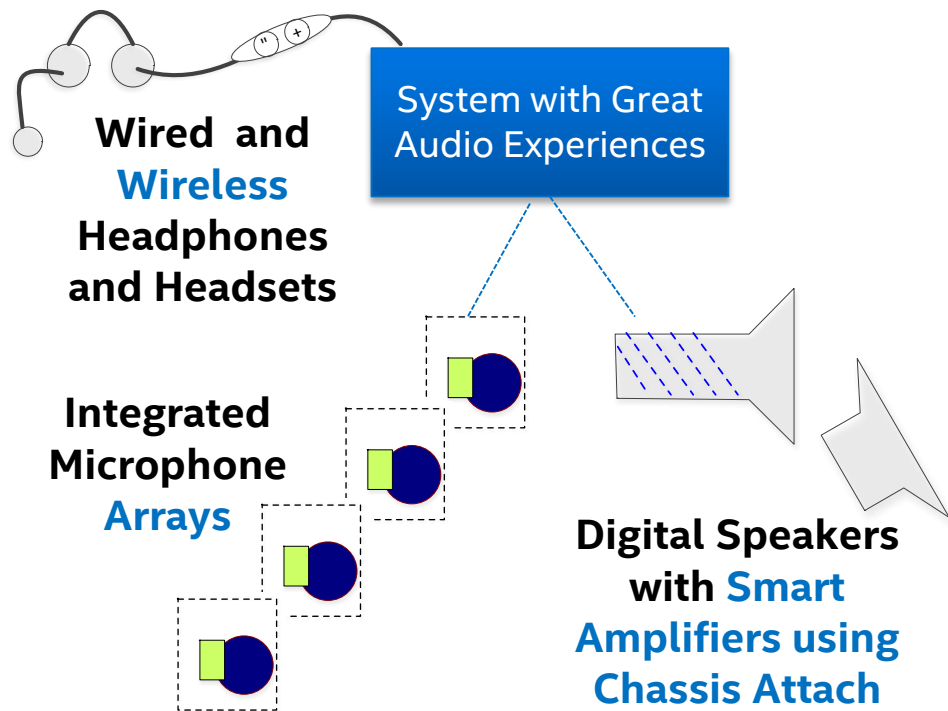
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# Audio Drives Key Experiences

- Interacting with **AI based bots and intelligent assistants** on your system or in the cloud
- **Communicating** with co-workers, friends and family from anywhere in the room **hands and eyes free**
- Enjoying **music** and **movies**



# Computing Renaissance

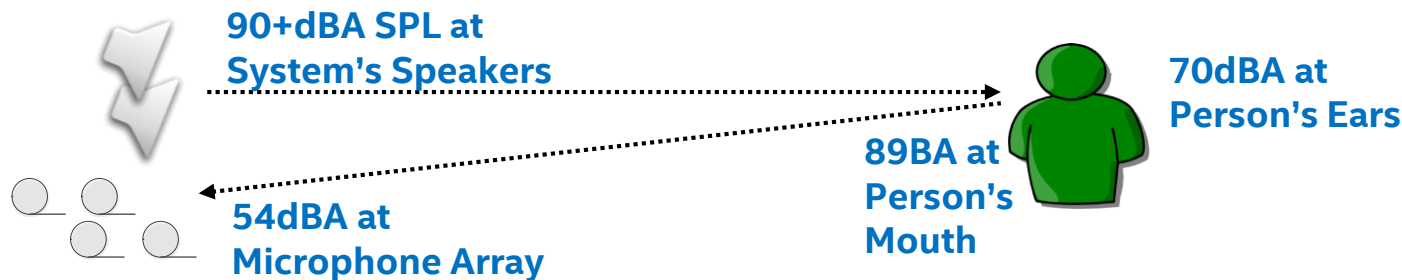
- We are entering a **computing renaissance**
  - Renaissance is being driven by speech (sometimes called voice) as new interface and new programming paradigm with AI based (chat) BOTs impact
  - AI and speech-based **assistants** and **bots** will have a dramatic impact on the man-machine interface
- Users will want to **use speech NOT text** with hands and **eyes free**
  - Speech outside of mobile becomes key capability
- The next generation of Intel CPUs and GPUs, wireless technologies, mobile screens and optics are going to make the man-machine interface richer and more immersive
- **Great microphones and speakers** critical for every computing device and system

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# What is “Far-Field Audio”

- “Far Field” is the ability to interact with system up to 4 meters away
  - Can be just in front of a system or full 360 degrees all around the device
  - Includes both clear voice calls and excellent speech recognition
- Transmission of human voice
  - Speech from 4 meters is only ~54dBA at microphones in reverberant rooms
  - Speaker output 85dBA is only 65dBA 4 meters away



- **Input and output** both must be good
  - Must be able to clearly hear device speech interaction responses far-end voice
  - Speech recognition challenge is to properly pick correct speech signal
- Requires improved algorithms, especially AEC and better speakers

# Wake on Voice for Far Field

- Far Field Speech is both **hands free** and **eyes free**
- Must be able to gain “**attention**” of the system without any physical contact
- Wake functionality provided by **key phrase** detection in low power states

20”(50cm)



100”(2 meters)



150” (4 meters)



Wake on Voice in Connected or  
Modern Standby

Wake on Voice in  
all low power states

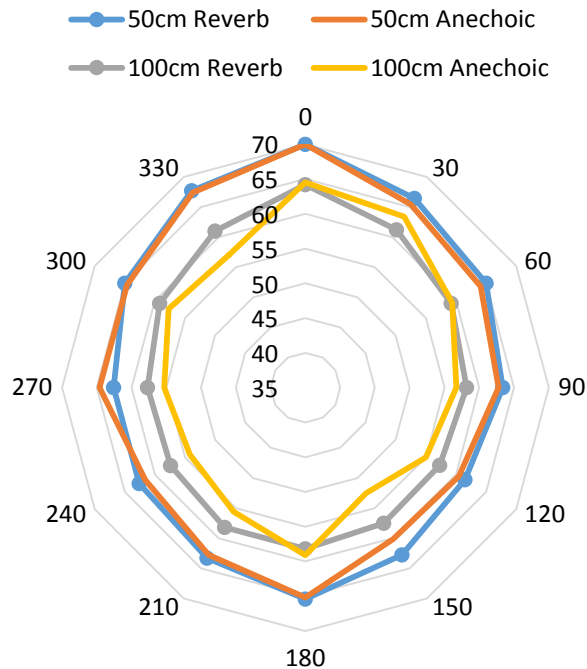
Expected for Future Speech Assistants



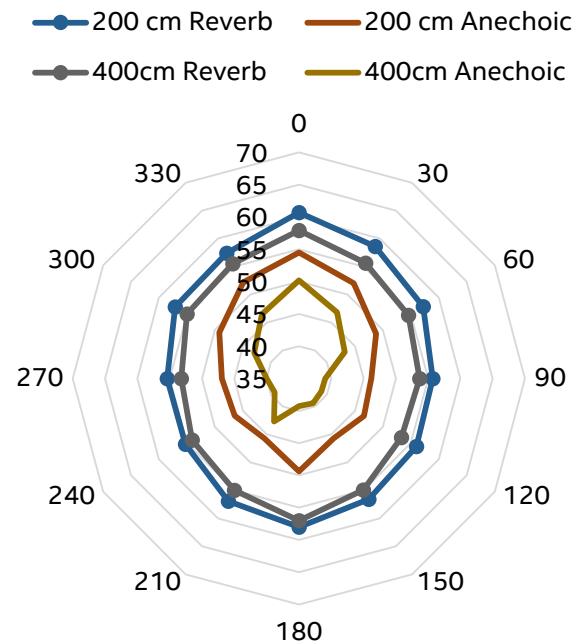
# Speaker Output and Reverberation

- Impact of reverb and speaker output

**SPL (dBA) around the notebook, with pink noise (reverb vs. anechoic room)**



**SPL (dBA) around the notebook, with pink noise (reverb vs. anechoic room)**



# Speaker Expectations for Far Field Usages

## Real world is what would be recreated for best experience

- Human & Machine Speech is 70-85dBa Average with at least 30dB of Dynamic Range attenuated for distance to listener although in noisy environments this might go as high as 90dB peaks; the Frequency for Speech is 100Hz to 16kHz
- Music listened to with at least 40dB of dynamic range and average levels about 70dB at your ears and thus must increase as you move away from the speakers with the frequency desire to go as low as 80Hz and at least up to 16-18kHz
- Movie sound track likely to have much higher crest factors with perhaps 80dB of dynamic range but with average listening levels again determined by the speech levels but adjusted in loudness as you go further from the speakers

For whole room (4 meters) real world best case desired

**100Hz-16kHz**  
**99dBa Peaks, 12-15dB crest**

**80Hz-20kHz**  
**88dBa Peaks, 18dB crest**

**80Hz-20kHz**  
**105dBa Peaks, 20+dB crest**

## Speaker Quality Metrics for Whole Room (Far Field) User Experiences

Basic Notebook or Convertible	Premium Notebook or Convertible	AIO, Portable AIO or "Amazon Echo* like System"
85dB	87dB	100dB
180Hz	165Hz	80Hz
MOS 3.7	MOS 4.1	MOS 4.5

*SPL is measured at 50cm distance and the Frequency at the "judged" -3dB Roll-off point*

# Great Speech when Using 4 Microphones

## High-Noise, High-Reverb Conditions

- ASR Engine: Cortana\* (CAR)
- Room Condition: RT60 ~450-500ms
- SNR: ~5dB
- Two **Non-Stationary** Noise Sources (TV Broadcast, Other Talker)
- User at 1m, 2m and 3m
- 120 Utterances; Combination of Male and Female Talkers



Interference



Interference

Conditions:	Unprocessed				4-MIC Processed		
	1m	2m	3m		1m	2m	3m
Speech Only	78.1	81.16	70.21		94.85	92.92	90.66
Speech + Noise	39.29	42.51	25.44		91.47	83.74	78.1
Speech + Noise + Echo	12.56	10.79	3.38		87.76	81.16	74.4

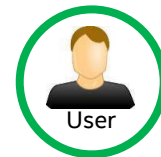
**In Low SNR Conditions, 4-MIC Technology Can Improve Performance up to 50% Compared to No Processing**



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# Designing for Speech Experience

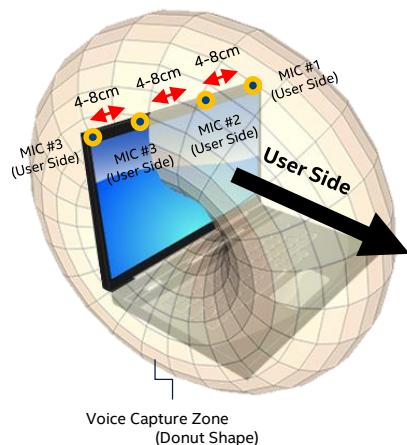
Design platforms that perform extremely well with seamless speech interaction and that works anywhere in the room, “Far Field”

- Using enough microphones with sufficient fidelity to provide 4 meter support
- Recommending **4 microphones** for most systems
- Use better quality microphones with **SNR of 63-65dB** or more
- No **singing CAPS**, HDD noise, **no fans or vents** near microphones, etc.
- Well **matched sensitivities** and phase across all microphones in an array
- Properly **sealing the mounting for the microphones** with good isolation for all internal system noises
- Best available signal processing algorithms **specifically designed for speech at 4 meters distances (Far Field)**
  - User and room noise position tracking with adaptive beam forming
  - Blind-Source Separation
  - Great Acoustic Echo Cancelation (AEC)
- Don't forget the **speaker output SPL** enough to be heard at 4 meters



# Far-Field Microphone Geometries

## 4-MIC Linear Placement



### MIC Placement

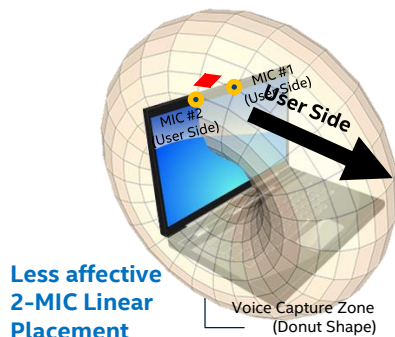
- Linear (user side), the four MIC can be placed anywhere on the same dimension

### Tracking Dimension

- 1-dimension

### Performance

- Better directive beam for far-field vs. 3-MIC
- World side noise will not be suppressed. Beam is formed in donut shape
- Donut size can expand to have 360° full range coverage



Less effective  
2-MIC Linear  
Placement

### MIC Placement

- Linear (user side), both MIC can be placed anywhere on the same dimension

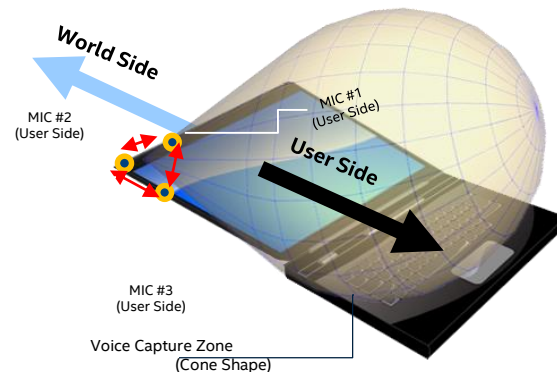
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## 3-MIC L-Shape Placement



### MIC Placement

- L-Shape (user side)

### Tracking Dimension

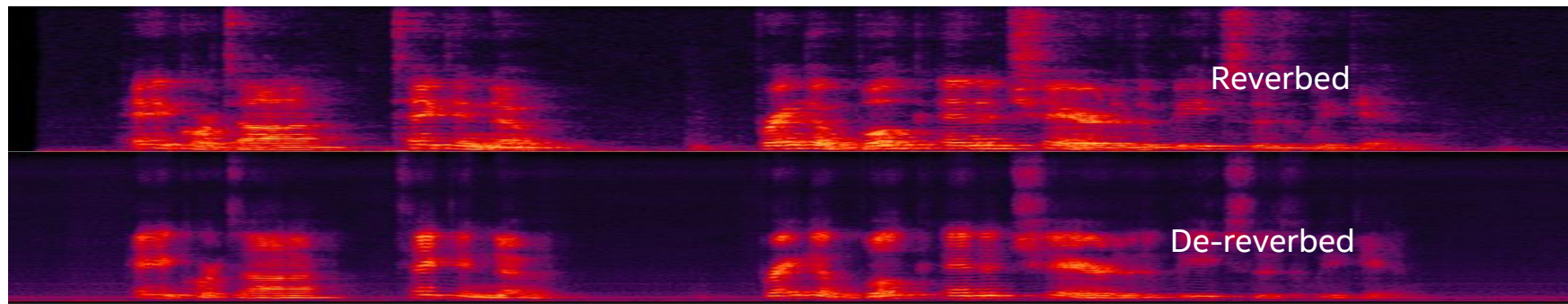
- 2-dimension

### Performance

- Better directive beam for far-field vs. 3-MIC
- World side noise will be suppressed when display is sitting "FLAT"

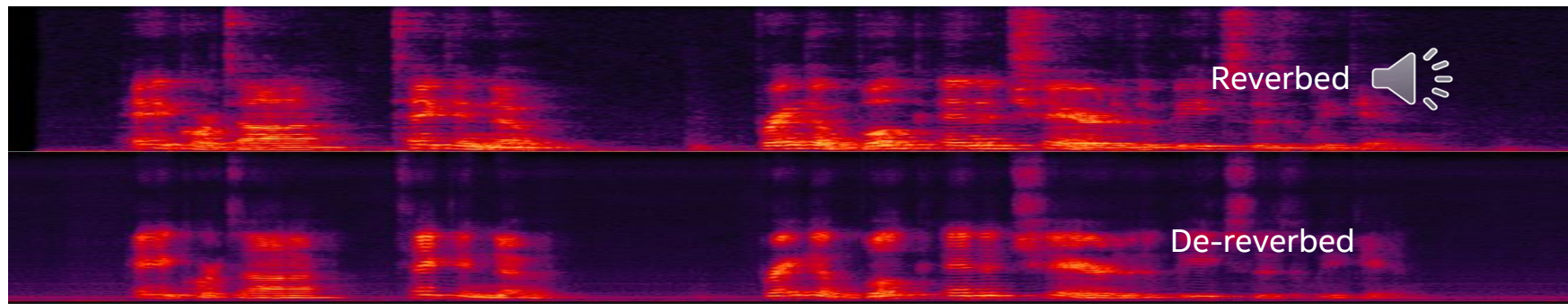
# Fortemedia\* Far-Field Speech Improvement

Success Rate (%)	3 meters					50 cm
	iS700* (15)	iS800V1* (2-Mic, 2Q16)	iS800V1* (4-Mic, 2Q16)	iS800V2* (2-Mic, 3Q16)	iS800V2* (4-Mic, 3Q16)	
Clean	68.26	73.21	75.12	84.83	90.85	98.01
Pub	54.71	67.81	69.18	76.17	82.88	95.12
Cafe	56.45	64.28	65.82	75.03	83.75	95.06
Music	36.35	55.26	62.22	67.61	73.15	96.32



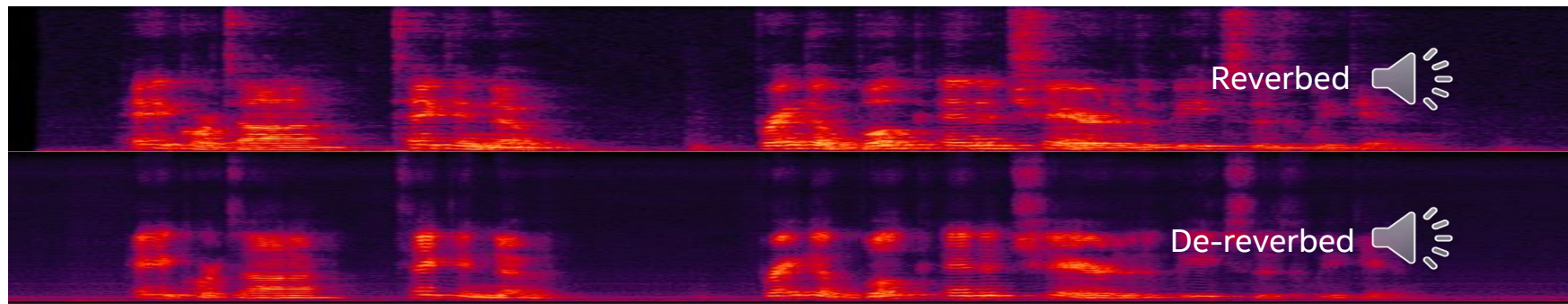
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# Speech Platform Evaluation Tool (SPET)

- Modular framework supporting multiple speech environments that validates platform speech subsystem and end to end user experience using form factor specific use cases
  - Not measuring specific ASR quality and performance
- Passing ensures good implementation of electromechanical microphone array design and algorithmic processing ensuring consistent speech recognition on Intel based platforms when using chosen speech environment such as Cortana\*, or RockHopper\*
- Validates the signal processing built in to the platform under various conditions including environment noise and platform orientation
- Provides advanced tools for supporting preprocessing tuning, batching data generation and performance analysis

# SPET Updates for 2017

- Adding additional tests specifically for key phrase recognition performance
  - Includes both false accept & false reject rate testing using multiple corpus
- Adding far-field microphone performance measurement support
- Includes room response measurements – reverb time (RT20/30/60), background NC/NR level, room impulse response
- Microphone frequency response, sensitivity
- Adds speaker frequency response, and loudness
- Improved equalization of background noise speakers at test unit

# Speech Assistant Requirement Expectations

	Today	Expected Next Generation Speech Assistant		
	Microsoft* Cortana* Standard	Standard	Premium	
Test Distance	0.5m	Near 0.5m	Near 0.8m	Far 4m
Noise Pattern	Quiet	Quiet		
	Babble_Pub	Babble_Pub		
	Babble_Cafe	Babble_Cafe		
	Echo	Echo		
		Babble_Pub + Echo		
		Babble_Cafe + Echo		
Speech Pattern	Male + Female	Male + Female + Children (5-12)		
Test Angle	0 deg	0 deg		
		50 deg		
Keyword Spotter	Only applicable for specific Windows* version	Yes		

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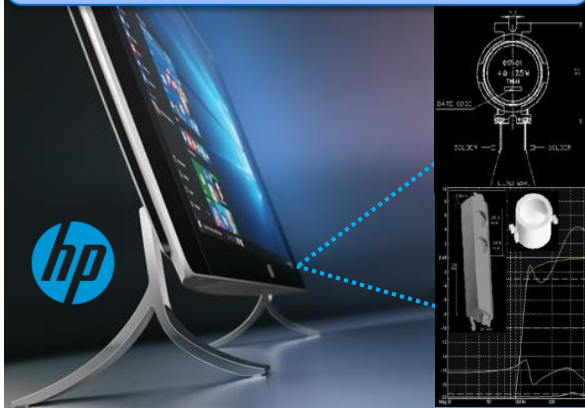
# Achieving Speaker Quality

Users feel that PC speaker quality not meeting expectations, matching competitive solutions and not ready for whole room and far field experiences

- A number of options to solve the challenges are available for you to use today...
  - Removing the speaker enclosure (the Box) and utilizing the chassis to provide the sealed volume called **Chassis Attached Transducer (CAT)**
    - Adds substantially to speaker volume (space)
    - Providing opportunity for reducing the audio BOM cost
  - **Smart amplifiers** driving enhanced peak power and providing **speaker protection**
  - **Larger speaker drivers** and use of either **dual magnetics** or dual drivers per side
  - **Digital and multi-coil speakers** with new amplifier enhancing loudness while lowering power consumption
  - **Front firing** of the speakers enhancing loudness and high frequency
  - Passive radiators helping to enhance bass

# Introduction to Sound Research

## Innovative Designs



- Patent-pending **technology**
- Optimize for **bass, loudness**
- System BOM **cost reduction**

**NEW** **Gasket Seal Technology**  
to complement CAT speakers

## Software



- Automated **speaker tuning**
- Best in class **enhancement**

**NEW** **Soft Amp**

**NEW** **CloudCal** room tuner

## Service

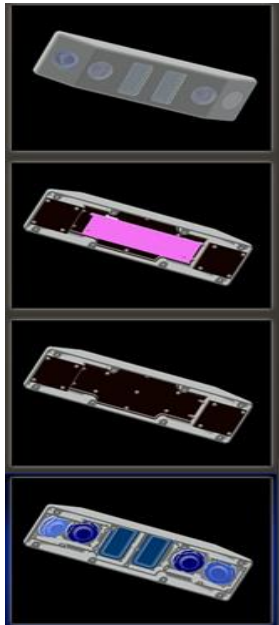


- Product **mastering services**
- Speaker **design consulting**
- OEM **training and support**

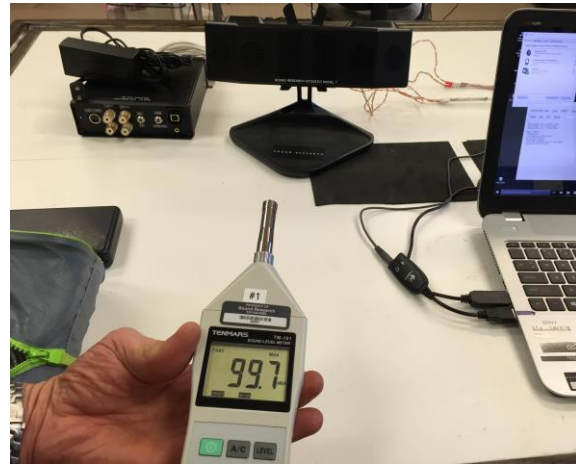
S O U N D   R E S E A R C H

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# The Smartest, Loudest Speaker is CAT



Hawking's voice was tuned by Stephen himself,  
using Sound Edge Tuning Tools



Hawking's Voice with  
CAT Speakers and  
Smart Amp  
99.7dB "A"

# Chassis Attached Transducers (CAT) Speakers and Gasket Seal Technology (GST)

- **Speakers require volume** (space) to achieve the best loudness and bass (low frequency)
- Thinner form-factors mean less internal volume to be utilized for speakers and mounting speaker boxes in a chassis means you're putting **a box inside of a box**
  - Plastic surrounding traditionally used, significantly reduces the usable volume
  - Speaker enclosure & mounting is complex & custom for every design adding to design & BOM cost
- CAT solution **removes the speaker enclosure (Box)** & **utilizes the chassis to provide the volume** while reducing the overall BOM cost
- Challenges with CAT:
  - High mechanical Q sealing
  - **Avoid glue** in platform assembly
  - ODM **HVM friendly**
- CAT without GST mounts directly to chassis using glue which easily cracks & flexes
  - Leads to distortion & poor bass response
  - Difficult & expensive to replace damaged or defective glue-attached transducers
- CAT with GST Speakers attach such that they are easily replaceable while eliminating separate fasteners/grommets where possible, reducing parts count and complexity
- GST supports speakers of all shapes & sizes



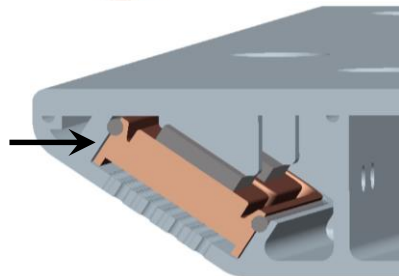
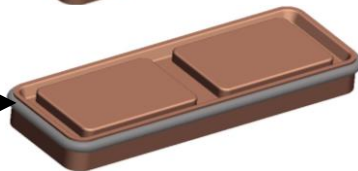
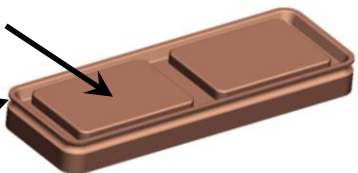
# Gasket Seal Technology

Speaker  
Transducer

Gasket  
Groove

Rubber  
Gasket

GST seals  
speaker  
into CAT  
enclosure  
built into  
chassis



## CAT/GST Advantages

- **Bigger Speaker Drivers**  
from more efficient use of space
- **Superior Bass**  
from additional back volume and better sealing (high "Q")
- **Maximum Loudness**  
from larger speakers and more stable attachment to chassis
- **Lower Support Cost**  
More reliable, easily replaceable
- **Potentially reduced BoM**  
Reduced parts count, less glue

## GST Partnership Speaker Vendors

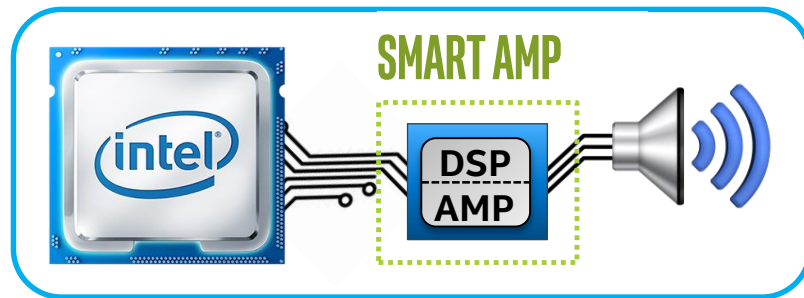


# Smart Amplifiers

## Another Approach for Loudness and Improved Bass

- Typically speakers are driven conservatively because of little or no direct knowledge of the overall system interactions (amplifier, speaker, enclosure, and chassis)
- Smart amplifiers can improve performance without compromising coil temperature, mechanical excursion, material fatigue, etc.

Smart...

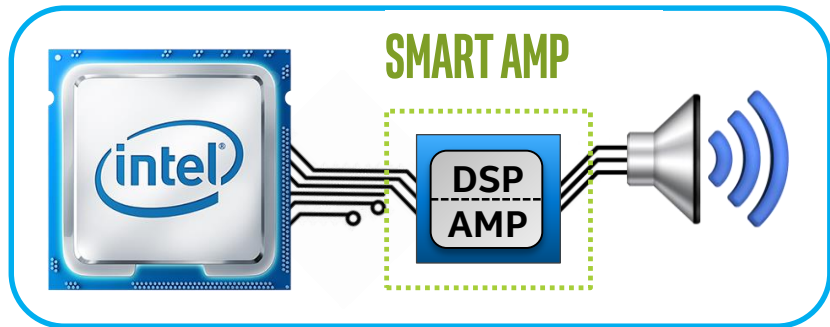


- Smart amplifiers utilize a separate DSP for processing to optimize the amplifier for the speaker you're driving
- Uses feedback mechanisms and known speaker characteristics from manufacturer to provide the best performance
- This means you are able to optimize the amplifier for the speaker in your system

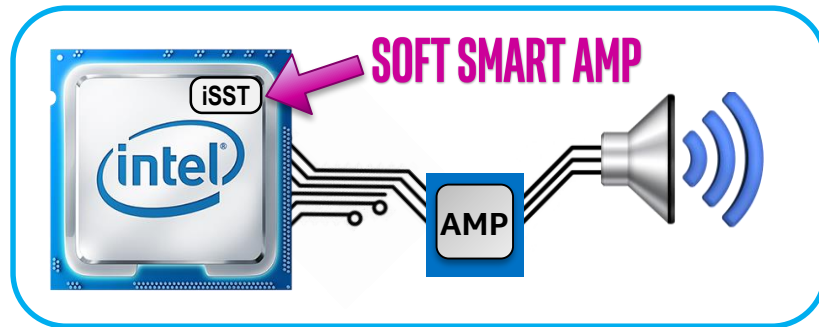


# Another Approach for Loudness

Smart...



Smarter...



**SOUNDEGE\***

**SOFT SMART AMP**  
**BENEFITS**

**CHEAP**

Potentially lower  
BoM with  
equivalent speaker  
driver protection

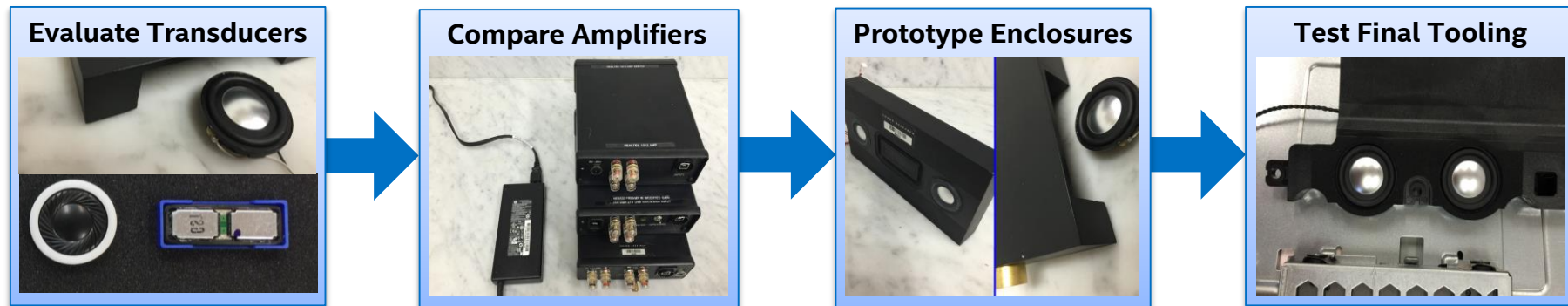
**LOUD**

Maximum SPL with  
minimal distortion  
and enhanced  
audio fidelity

**GOOD**

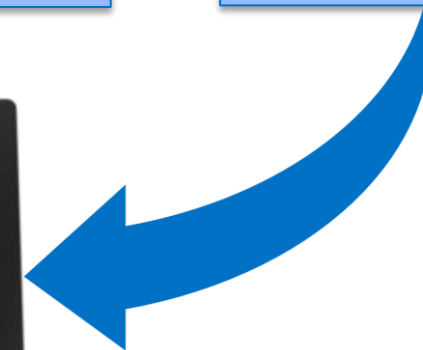
Better sounding  
solution when  
tuned with  
SpeakerCal\*

# Example with 2016 HP Envy\* All-In-Ones



## Design Goals

- Maximize **SPL**
- Optimize **Bass**
- Reduce **BoM Cost**



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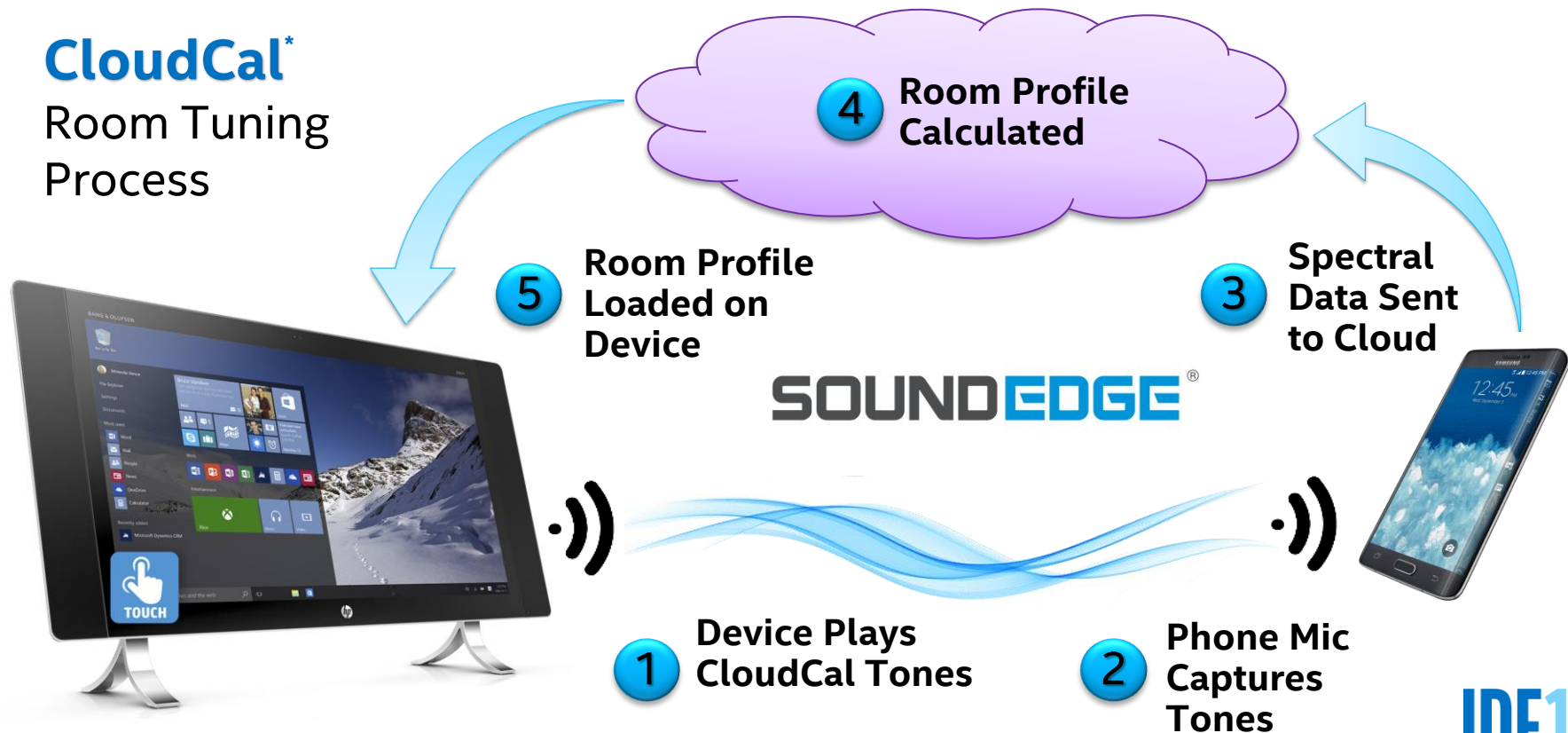
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# Software Example: CloudCal\* Room Tuner

**CloudCal\***

Room Tuning  
Process



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# Innovation with Audio

## Key User Experiences

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Conferencing and speech recognition from **anywhere in the room**

**Great sounding speakers** so don't have to use headphones

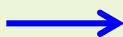
## Ready Today



Great microphones, algorithms and new improved speech engines such as Cortana\* and Amazon\* Alexa\*



Improved MEMS microphones and noise reduction algorithms including source separation



New amplifiers and speaker driver technologies such as multi-coil, speaker modules and enclosure-less

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## New User Experiences

**VR with fantastic gaming, conferencing spatial audio** with head tracking

Audio in **multiple rooms** simultaneously

User's **identification and biometrics** using voice and speech

Continuously **sensing** and adapting to environment

**World and user** side audio for conferencing and content creation

## Opportunities

- Digital headphones with 6-axis reporting and new low latency and power algorithms
- Synchronization primitives, network streaming
- Integrated DSP with low power new biometrics algorithms
- Multi-modality, open DSP, low power microphone, algorithms and apps
- Microphone arrays with world facing microphone, new algorithms with input from 3/6 axis sensors providing spatial audio

# Intel® Smart Sound Technology Evolution

Today

## Low-Power Audio, Voice & Speech



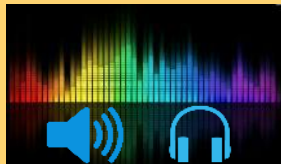
Low-power, end-to-end audio, voice and speech processing

## Longer Battery Life



Integrated Audio DSP enables new user experiences and saves power

## High-Quality Audio Fidelity



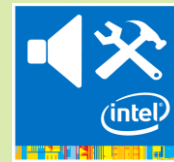
Optimized speakers and headphones performance takes PC audio to next level

## Intel® Wake on Voice



Wake up your PC and use voice for control and assistance (including in standby)

## Open DSP FDK



Developer access to Audio DSP to create new experiences

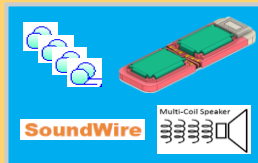
Next Generation

## Enhanced Audio DSP



2x performance and 50% more memory to enable more advanced features

## Audio Platform Innovations



Simplified interface, improved audio/speech fidelity, and reduced power

## Improved Speech Recognition



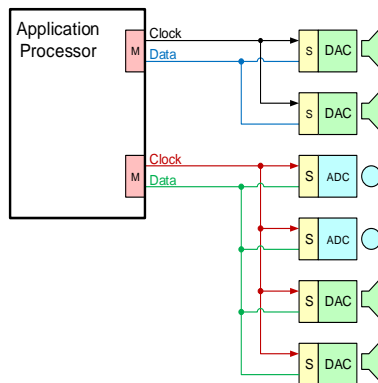
Hardware acceleration for faster and more accurate speech recognition

All information provided here is subject to change without notice. Contact your Intel representative to obtain the latest Intel roadmaps.

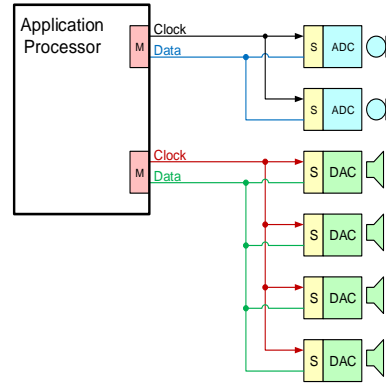
# SoundWire\*

- Robust, scalable, two-pin multi-drop bus
- Low power, low latency, well-bounded PHY and transport with master and slaves
- Embedded control, commands, status
- In-band interrupts, wakes with clock stop
- Multiple dynamically reconfigurable streams
- PCM, PDM, isochronous and asynchronous transfer modes
- Fast device configuration up to 20 Mbits/s
- Multiple clock frequencies up to 12MHz
- Double data rate signaling, lower power

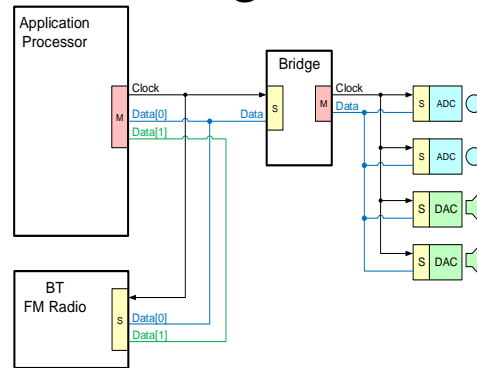
## Functional Partitioning



## Routing/Use Case Partitioning

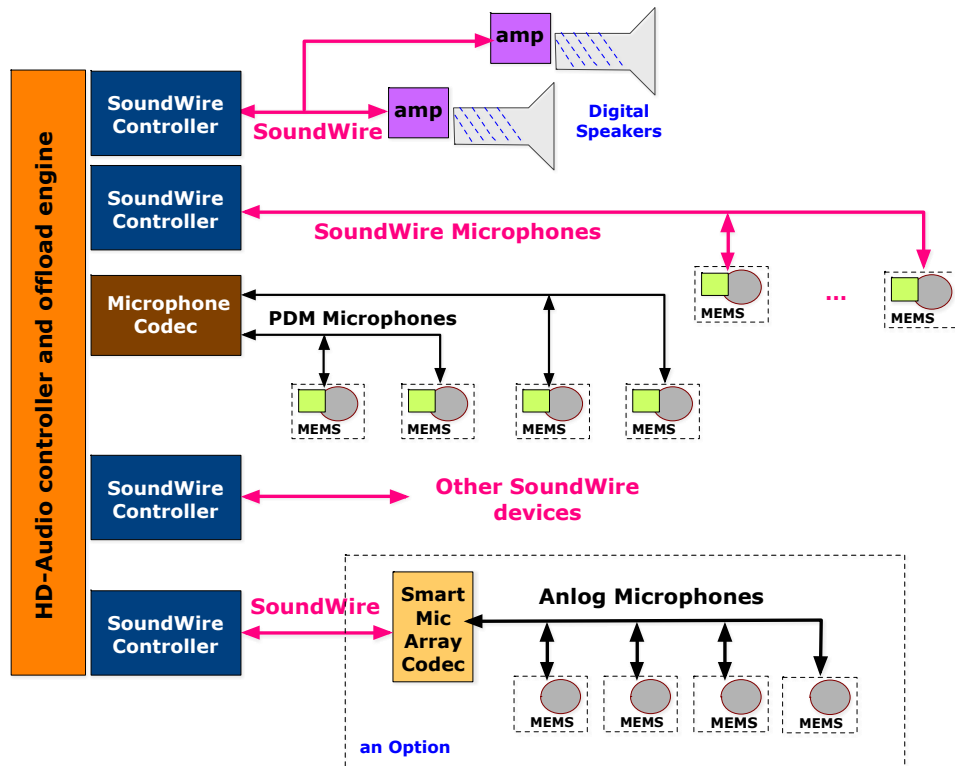


## Bridged



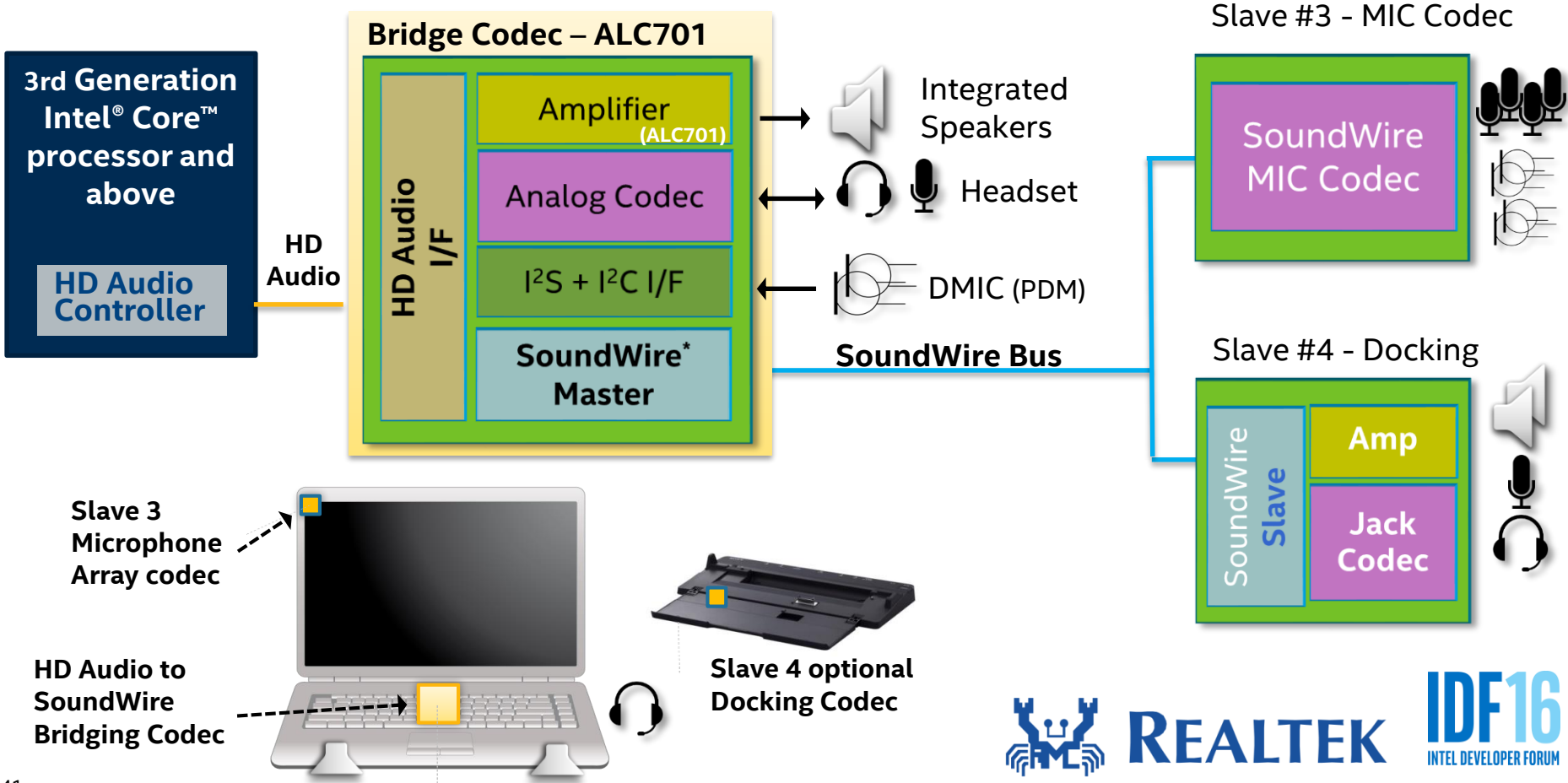
# SoundWire\* Integration

- Four controllers each with their own physical link (segment)
  - Two physical interfaces are shared with DMICs
  - 3<sup>rd</sup> and 4<sup>th</sup> take back some pins from I2S & HD-audio
- SoundWire\* interfaces are part of HD-audio controller
- SoundWire support will require the use of the audio offload engine (a hardware DSP) & the bus and offload drivers as part of Intel® Smart Sound Technology

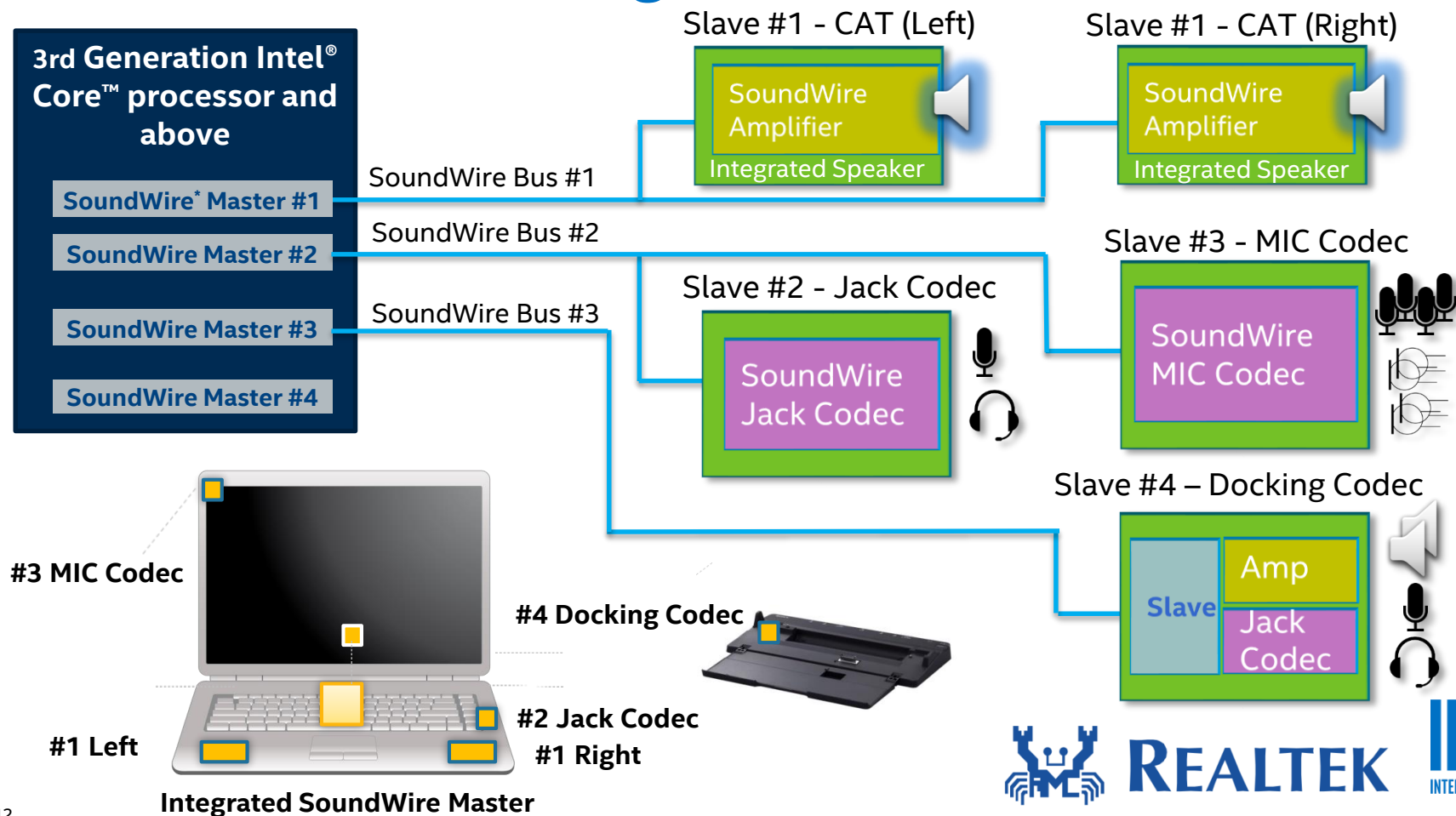




# Bridging Architecture Available Now



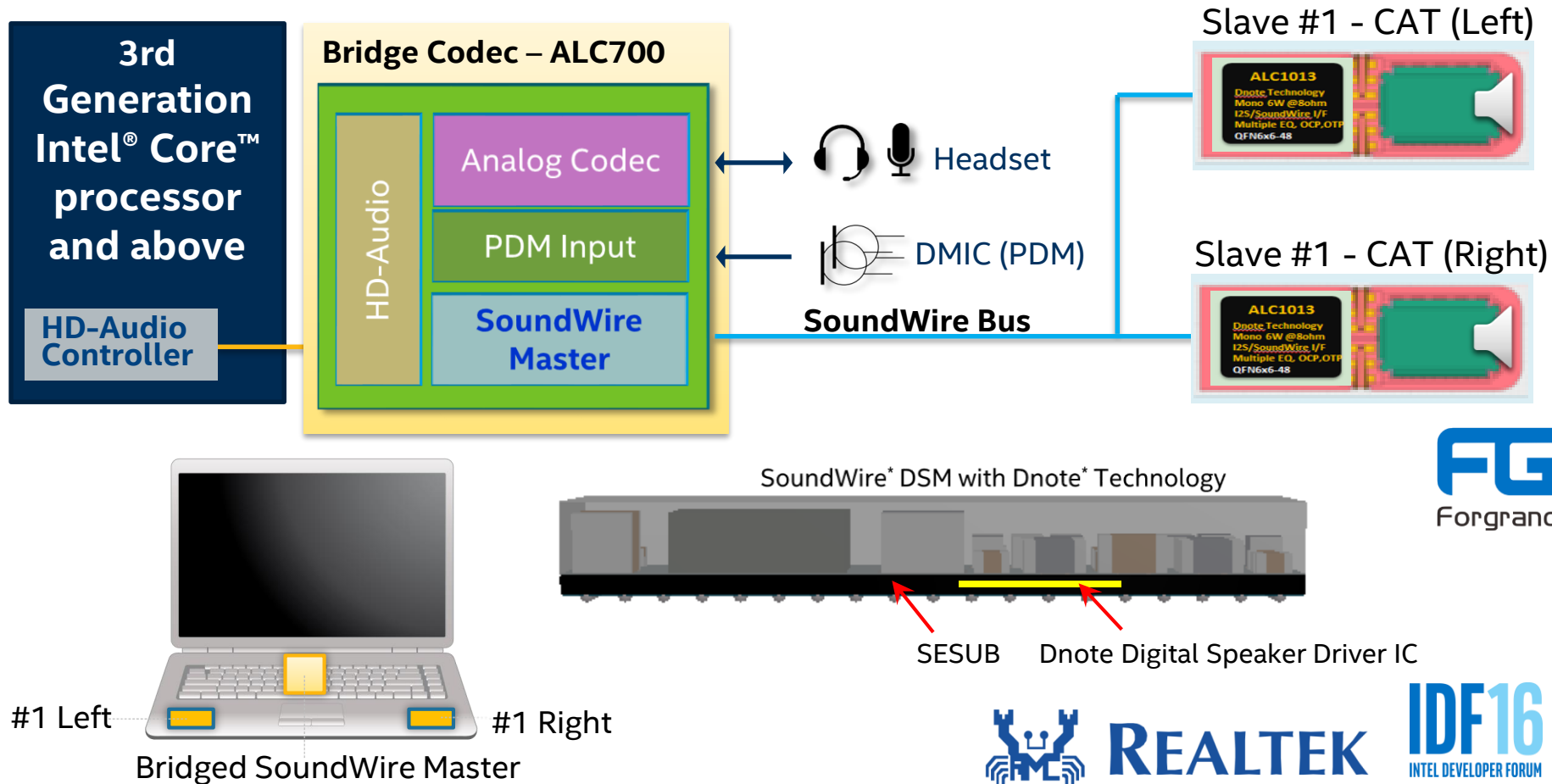
# SoundWire\* with Integrated Controllers



REALTEK

IDF16  
INTEL DEVELOPER FORUM

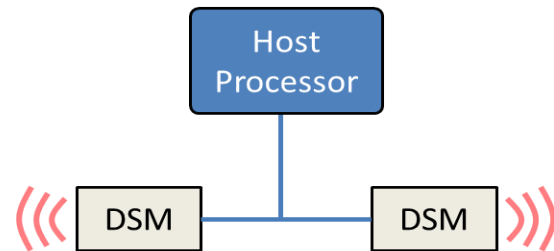
# SoundWire\* Digital Speaker Module (DSM)



# Digital Speakers and Modules (DSMs) Using Trigence\* Dnote\* Technology

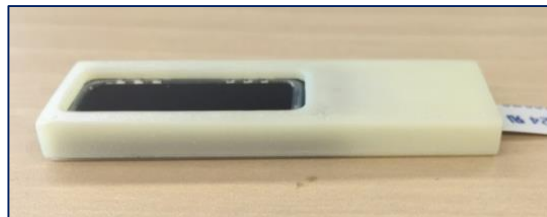
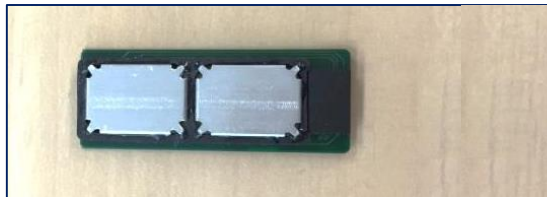
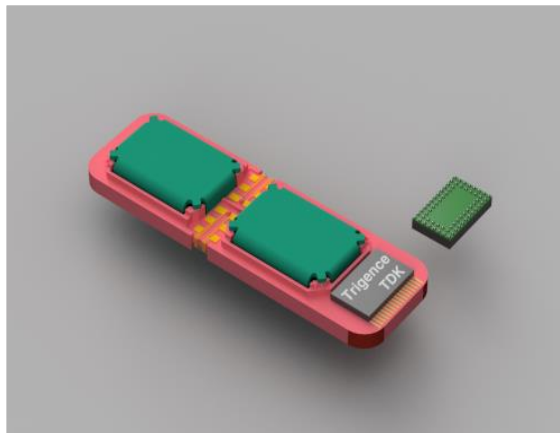


- Self-contained digital components
- Complete digital speaker sub-systems
- Individually addressable at the host level
- Incorporate a bi-directional digital audio interface, audio DSP, Dnote\* processor, and digital speaker within a single speaker from

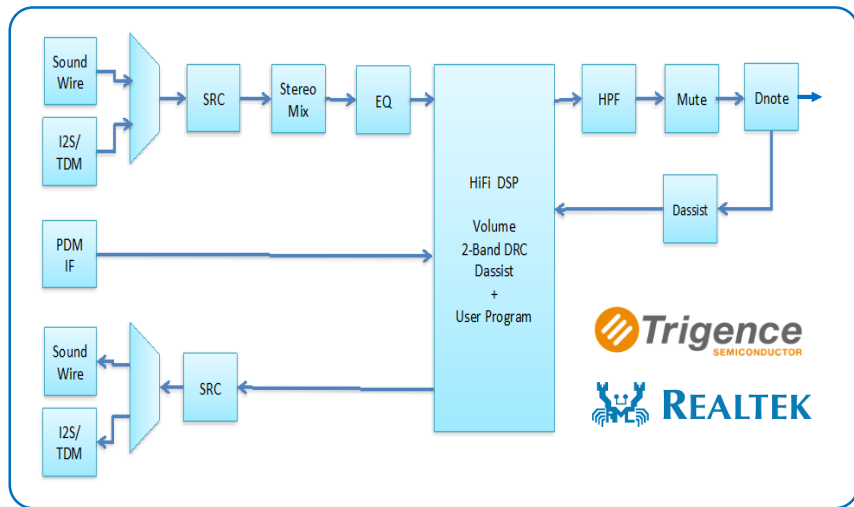


## Advantages

- Lower Power and Supplies
- Improve Audio Quality
- Simplify System Architecture
- Offer Advanced Functionality
- Lower Overall Cost
- Faster design and qualification



# Digital Speaker Driver IC (ALC1013)



- **SoundWire\***, I<sup>2</sup>S and TDM audio interfaces
- 200 MHz Tensilica\* HiFi2 DSP (~100MHz available for customer IP / functionality)
- **D-Assist** speaker compensation algorithms - derived from active measurement of speaker characteristics

- **Dnote\* Digital Amplifier System**
- True 24-bit audio output with < -120dB in-band noise floor
- Greater output power than a conventional Class-D amplifier from the same power supply voltage

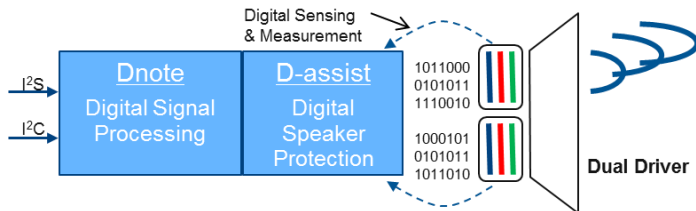
Power Supply	Dnote (8Ω x 6)	Class-D (8Ω)
1.8V	0.75W	0.15W
3.3V	2.55W	0.45W
5.0V	6.75W	1.0W

- Improved Power Efficiency (1W into speaker)
  - Dnote consumes **1.1W from system**
  - Typical D-Class consumes **2.1W**
- All digital / generic bulk CMOS solution



# Speaker Compensation with D-Assist\*

- D-Assist\* is a suite of speaker compensation algorithms designed to maximize the performance of digital micro-speakers through the precise sensing and compensation of physical electro-mechanical properties
- D-Assist makes use of techniques that are only possible through the close coupling of the driving and sensing electronics with the speaker mechanics



## DSA - Digital Smart-Amp

- Entirely digital smart-amp implementation
- Offers speaker protection and greater perceived SPL
- Does not rely on current sensing resistors or probe-tones
- No loss in efficiency over non smart-amp implementation

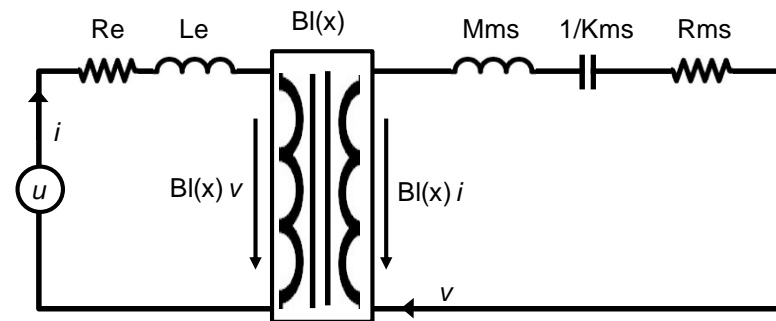
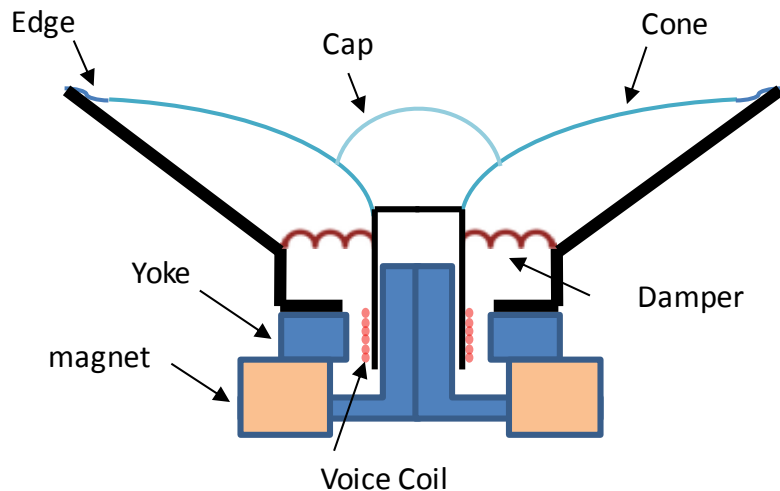
## DDR - Digital Distortion Reduction

- Dynamically compensates THD (Total Harmonic Distortion) degradation in the speaker
- Typical overall speaker THD reduction to less than 20% of uncorrected value

## DTSC - Digital Thiele-Small Correction

- Uses active measurement of speaker characteristics to provides control over and compensation of a speaker's Thiele-Small Parameters through use of on-chip DSP

# Thiele-Small Parameters



**$R_e$**  : Coil Resistance

**$L_e$**  : Coil Inductance

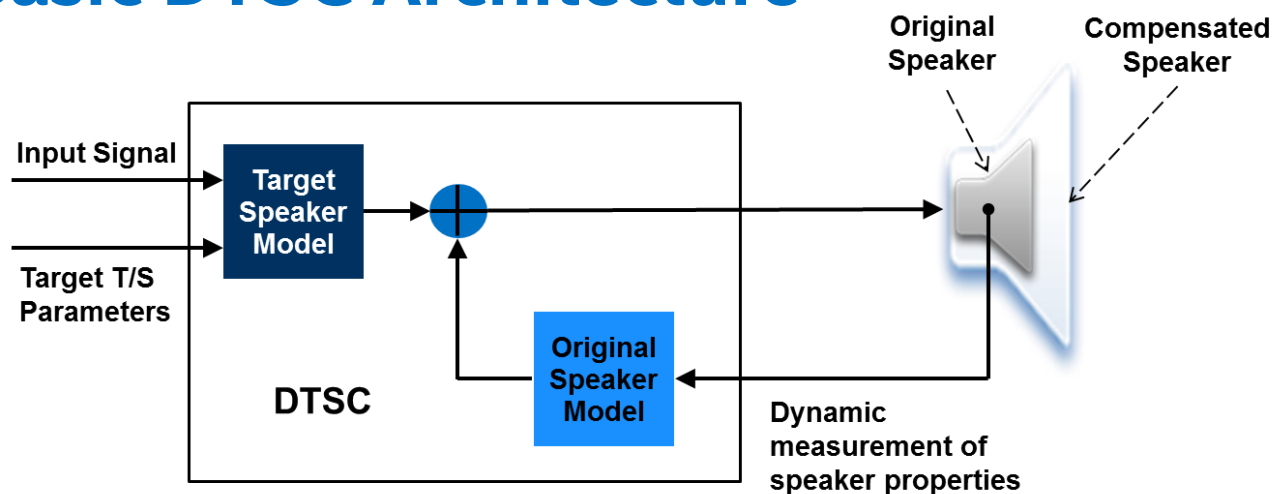
**$BI$**  : Force Factor

**$K_{ms}$**  : Mechanical Stiffness

**$R_{ms}$**  : Mechanical Resistance

**$M_{ms}$**  : Driver Mass

# Basic DTSC Architecture



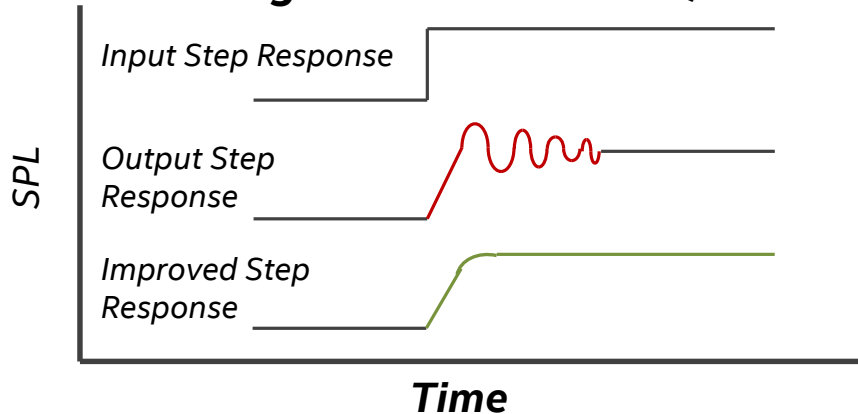
- Uses physical model of transducer and modern feed-forward control theory to permit the **time-domain equalization** of the speaker's transient response
- Original Speaker approximates the same sound as Target Speaker model if measurement of Original Speaker parameters is accurate





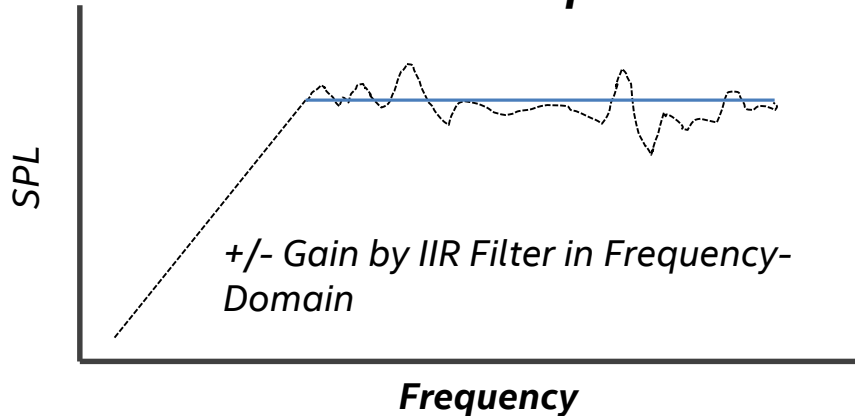
# Comparison with Frequency Domain EQ

## Trigence\* reSound\* EQ



- Can control signal response
- No group delay or phase shifts
- Minimal THD or noise degradation
- No need for “golden ear” to perfect sound
- **Easy** to perceive improvement

## Conventional Equalizer

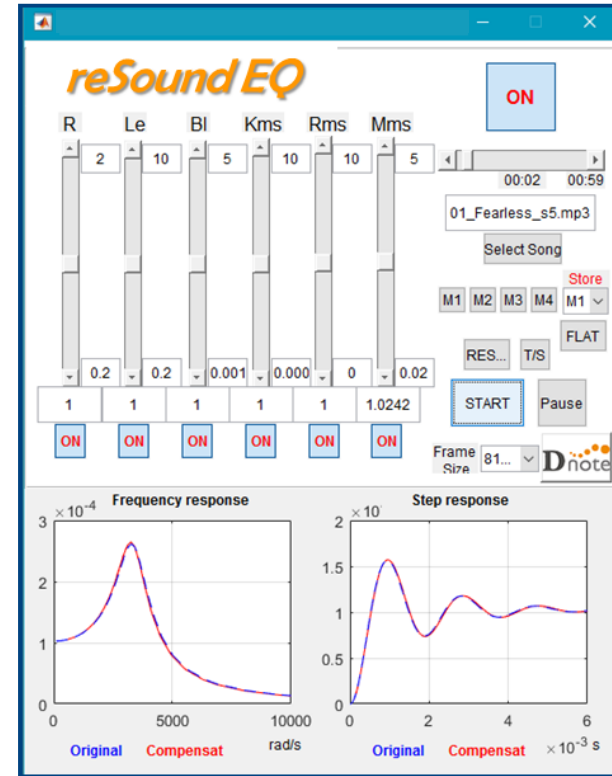


- Cannot control signal response
- Inherent phase shift /delay through IIR filters
- Needs “golden ear” to perfect sound flavor
- Degrades THD and increases noise level
- **Difficult** to understand improvement or not

# Trigence\* reSound\* EQ

- Works with all ALC1013-based Digital Speaker Modules (DSMs)
- Enables dynamic tuning of DTSC functionality within the speaker
- Fast, simple speaker alignment / compensation tool
- Designed specifically to improve Digital Speaker transducer quality

## reSound EQ demo



# Agenda

- Introduction
- Far Field Voice and Speech
- Output and Speakers
- Innovation, SoundWire and New Audio Hardware
- **Wrap Up**

# Summary and Next Steps

- Computing renaissance being driven by audio especially voice and speech with AI and bots, thus becoming critical to competitiveness
- Far field for at least 4 meters voice and speech can be supported today with the new generation of improved processing algorithms from multiple vendors, improved microphone arrays with more microphones
- Speakers with 85dB SPL and 180Hz can be provided today using larger drivers, digital speakers, improved amplifiers and CAT
- SoundWire\* and new audio hardware should be adopted where possible

# Additional Sources of Information

- A PDF of this presentation is available from our Technical Session Catalog: [www.intel.com/idfsessionsSF](http://www.intel.com/idfsessionsSF). This URL is also printed on the top of Session Agenda Pages in the Pocket Guide.
- See Realtek\*, Trigenice\*, Forte Media\* and Conexant\* in the showcase

# Technical Sessions in PC Innovation Track

## Tuesday, August 16, 2016

**11:00 AM – 12:00 PM PCITS02** — Designing to Meet New Expectations for Audio Experiences and Standards Critical for Competitiveness **Level 2 Room 2009**

**11:00 AM – 12:00 PM VRGTS01** — Bringing Virtual Reality to the Mainstream **Level 2 Room 2005**

**1:15 PM – 2:15 PM VRGTS02** — Overclocking Innovation: 2016 Enhancements and New Opportunities **Level 2 Room 2005**

**2:30 PM – 3:30 PM SOFTS03** — ChromeOS\* and coreboot\* on Intel® Architecture Platforms – A Primer **Level 2 Room 2006**

**2:30 PM – 3:30 PM VRGTS03** — Developing Virtual Reality Solution with Intel® Architecture **Level 2 Room 2005**

**4:00 PM – 5:00 PM CONTS02** — Simplified Platform Power Measurement Using USB Type-C\* Interface to Drive Software Power Optimization **Level 2 Room 2008**

**4:00 PM – 5:00 PM SOFTS05** — Intel® Software Guard Extensions Technology Overview and Programming Model **Level 2 Room 2006**

## Wednesday, August 17, 2016

**11:00 AM – 12:00 PM PCITS01** — Touch/Stylus Technologies for 2016 – Intel® and Ecosystem Drives Improved Touch/Stylus User Experience **Level 2 Room 2000**

**1:15 PM – 2:15 PM CONBZ03** — Thunderbolt™ 3 Technology – The USB Type-C\* that Does It All **Level 2 Room 2007**

**1:15 PM – 2:15 PM PCITS03** — Google Play\* on Chrome OS\* + Intel® Architecture – A Primer on Developing the Best Apps **Level 2 Room 2000**

**1:15 PM – 2:15 PM SOFTS06** — Machine Learning: Optimizing Deep Learning Usages on Intel® Client Platform **Level 2 Room 2006**

**2:30 PM – 3:30 PM CONTS03** — Shape the Personal Home Gateway: Flexible Design for the Connected Home **Level 2 Room 2007**

**2:30 PM – 3:30 PM PCITS04** — Evolving the 2 in 1 through New Usages, Technologies, and Form Factor Design Optimization **Level 2 Room 2000**

**4:00 PM – 5:00 PM PCITS05** — Intel® Authenticate – Innovation and Multi-Factor Authentication for Enterprise **Level 2 Room 2000**

## Thursday, August 18, 2016

**9:30 AM – 10:30 AM MASTS06** — Unifying the Storage Interface Across Client and Data Center **Level 2 Room 2002**

**9:30 AM – 10:30 AM PCITS06** — Enhancing and Extending Intel® Unite™ with Plugins **Level 2 Room 2004**

**10:45 AM – 11:45 AM PCITS07** — Driving Enterprise Value with Intel® Core™ vPro™ Processor **Level 2 Room 2004**

**2:15 PM – 3:15 PM PCITS08** — Modern Standby: Why and How **Level 2 Room 2004**

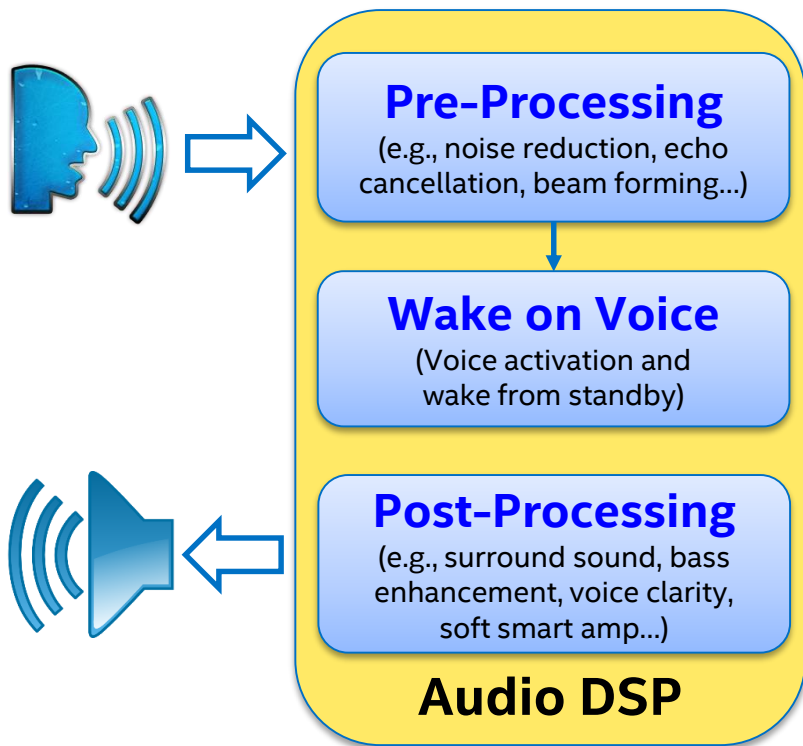
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# Backup



# Intel® Smart Sound Technology Audio DSP



- **Enhanced voice, speech and audio quality** from microphone(s)
- **Improved voice communications and speech recognition accuracy**



- Supports **voice activation from S0** and **wake from standby**
- **Improved battery life** running on audio DSP -vs- host CPU

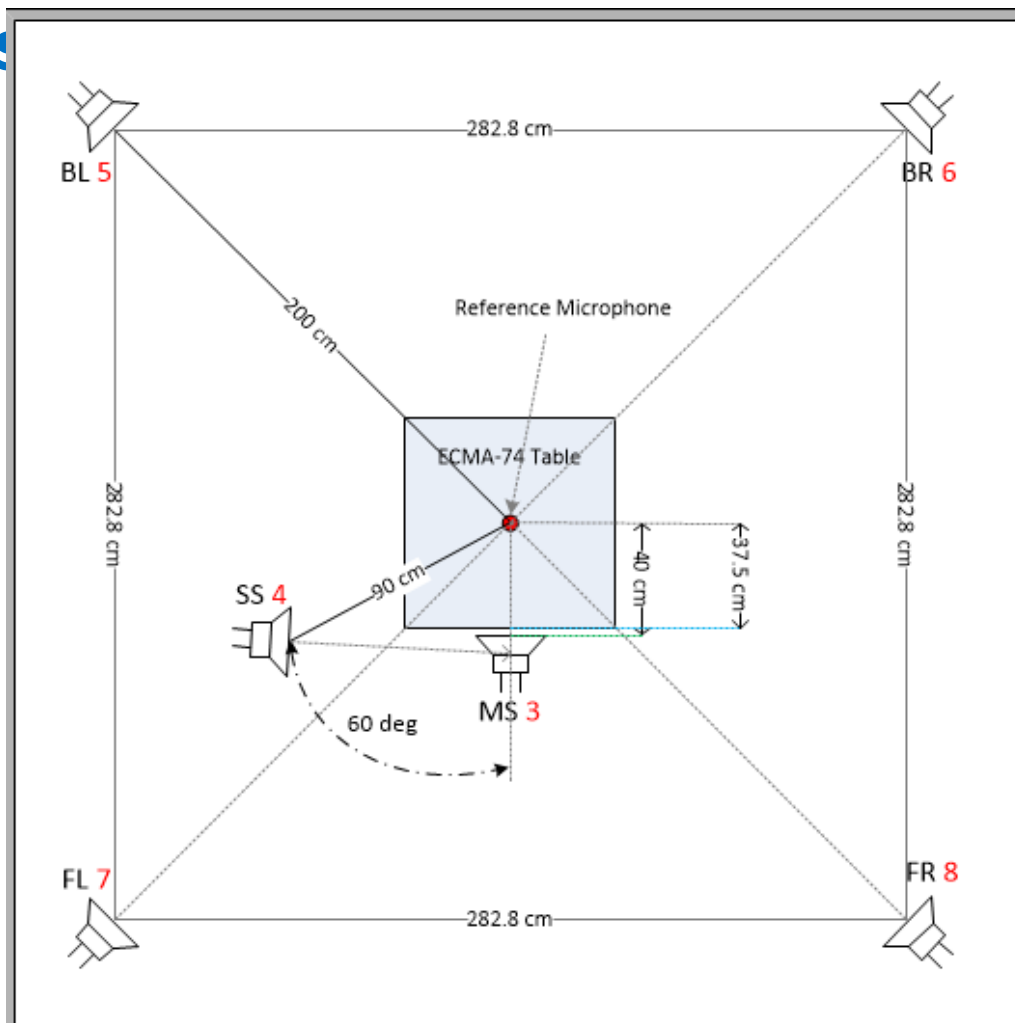
AUDYSSEY



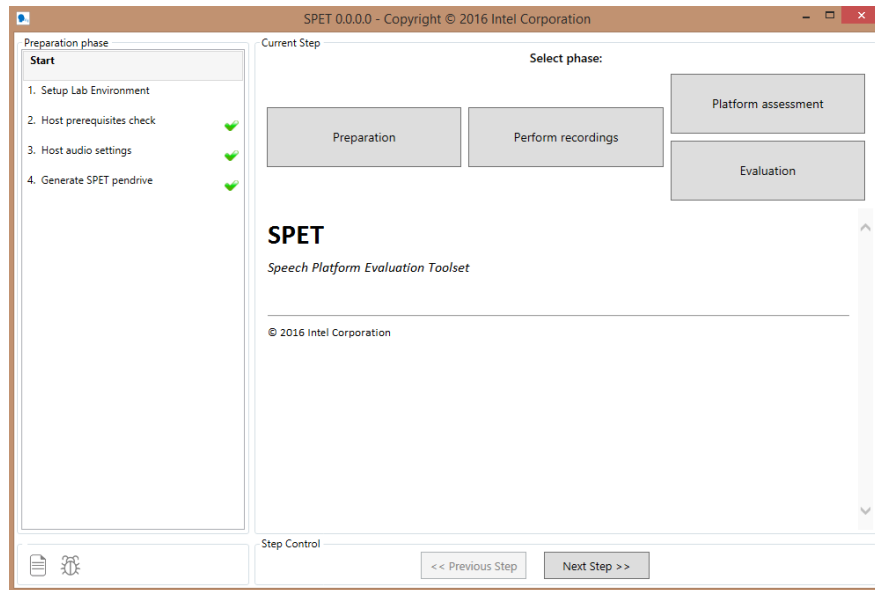
- **Enhanced audio playback quality** for headphones and speakers
- Soft smart amp for **cost-reduction, speaker protection and improved audio fidelity**

# SPET Updates for 2017

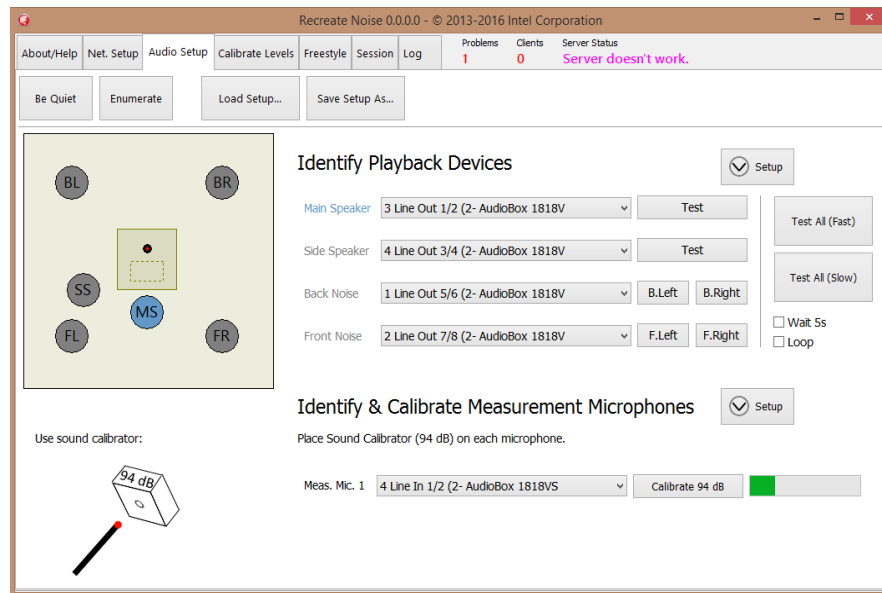
# SPET Updates



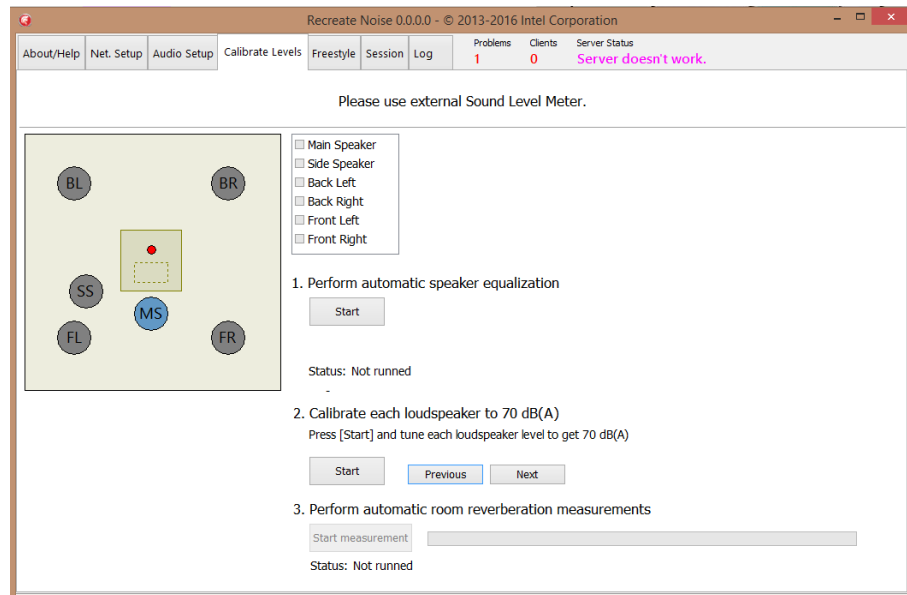
# SPET Updates for 2017



# SPET Updates for 2017



# SPET Updates for 2017



# SPET Updates for 2017

- ☐ Main Speaker
- ☒ Side Speaker
- ☐ Back Left
- ☐ Back Right
- ☐ Front Left
- ☐ Front Right

## 1. Perform automatic speaker equalization

Stop

Total progress:



Recording Side Speaker [attempt 1/4]

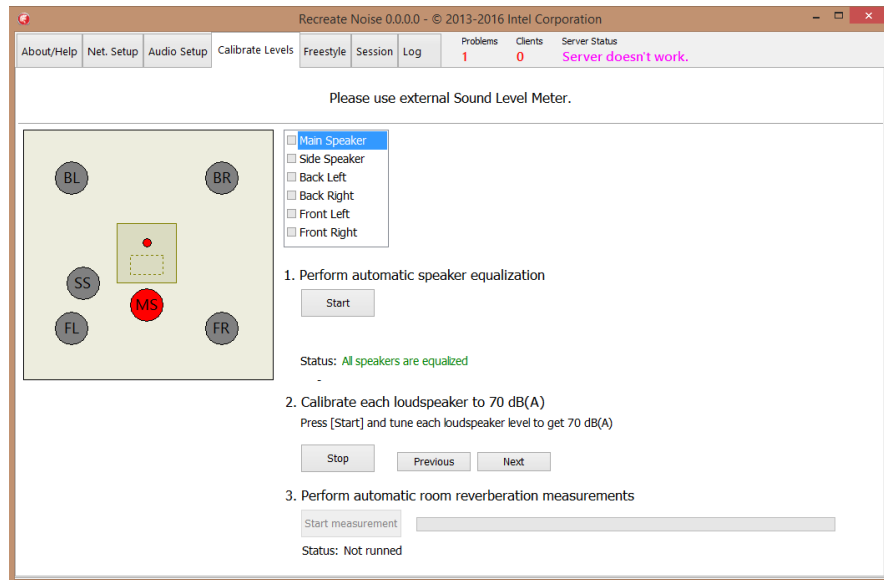


Status: Running...

# SPET Updates for 2017



# SPET Updates for 2017



# SPET Updates for 2017

SPET 0.0.0.0 - Copyright © 2016 Intel Corporation

Perform recordings phase

Start

1. Device description ✓
- 2. Select recordings ?**
3. Setup DUT computer ?
4. Perform recordings ?
5. Transport files from DUT ?

Current Step

### Recording session

Available sessions: Platform assessment ▼

Noise conditions

☐ Close field:

☐ Far field:

Reverberation level: ☐ Low ☒ Medium ☐ High

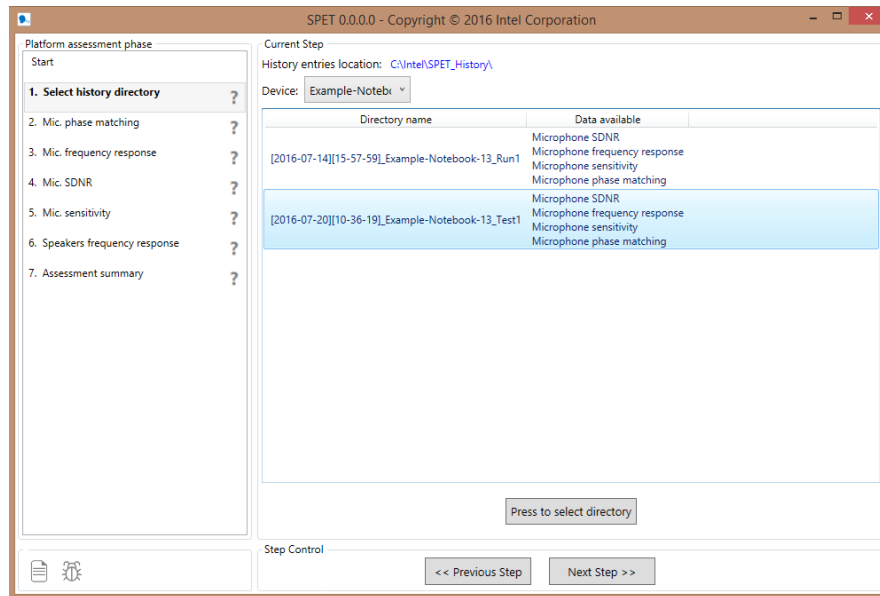
Short recording description (e.g. 'RAW', 'XYZ Effects'):

Step Control

<< Previous Step

Next Step >>

# SPET Updates for 2017



# SPET Updates for 2017

The screenshot displays the SPET 0.0.0.0 application window. On the left, the 'Evaluation phase' sidebar lists four steps: 'Start', '1. Select history directory', '2. Evaluation session settings', and '3. Calculation' (which is highlighted with a question mark icon). Below this is '4. Evaluation report'. The main area is titled 'Current Step' and shows the following details:

- History entry: [2016-07-20][10-36-19]\_Example-Notebook-13\_WOV2
- Evaluated speech length: WovShort (0h:20m)
- ASR plugin used: Intel WOV
- Estimated computation time: 00:04:00 (Note: The actual time will vary depending on HOST computer specification)
- Actual last computation time: 00:05:13

There are two checkboxes on the right: 'Skip WAV synchronization' and 'Keep temporary data'. A 'Start FAR/FRR calculation' button is located below the timing information.

Condition	Clean	Cafeteria	SideSpeaker	MusicEcho
RAW	FRR score: 84%   FA/h scor	FRR score: 92%   FA/h scor	FRR score: 85%   FA/h scor	FRR score: 51%   FA/h scor

Below the table is an 'Evaluation log' section with a dropdown arrow and a 'Copy results to clipboard' link.

At the bottom, the 'Step Control' section includes '<< Previous Step' and 'Next Step >>' buttons.

# Transducers Technology Roadmap

Before

Now

Analog

PDM

I2S

SoundWire\*

---

# Transducers Technology Roadmap

Before

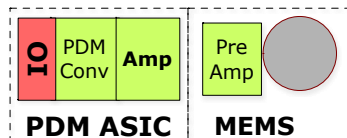
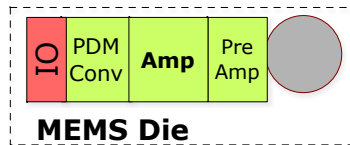
Now

Analog

PDM

I2S

SoundWire\*



Arrays

Improved Mounting

# Transducers Technology Roadmap

Before

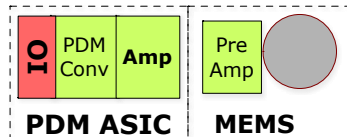
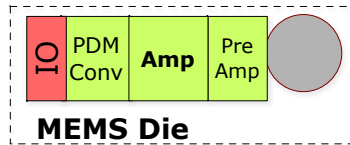
Now

Analog

PDM

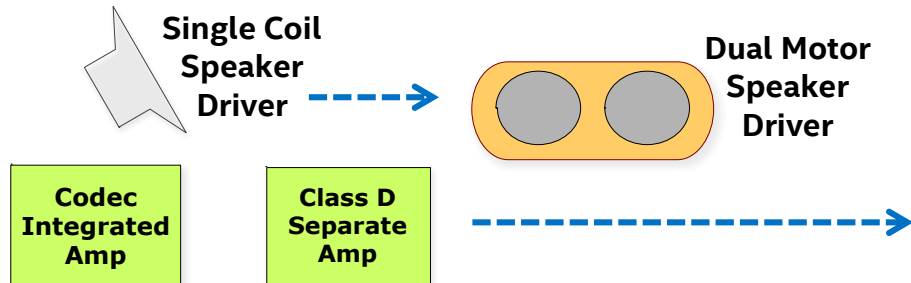
I2S

SoundWire\*



Arrays

Improved Mounting



# Transducers Technology Roadmap

Before

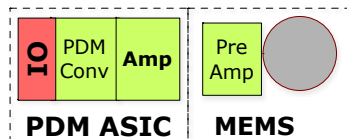
Now

Analog

PDM

I2S

SoundWire\*



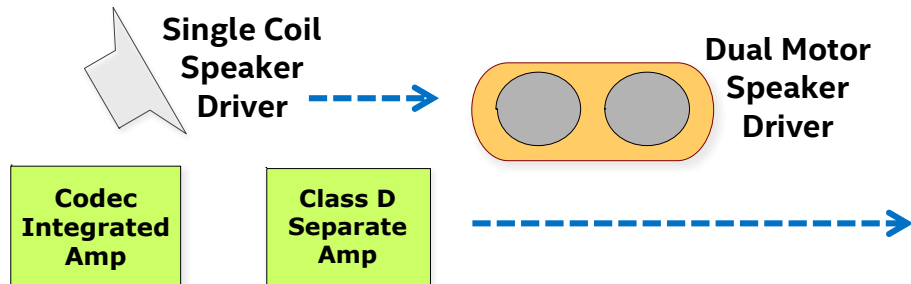
Sensing

Controllable Gain

Arrays

Improved Mounting

Ultrasonics





# Transducers Technology Roadmap

Before

Now

Analog

PDM

I2S

SoundWire\*



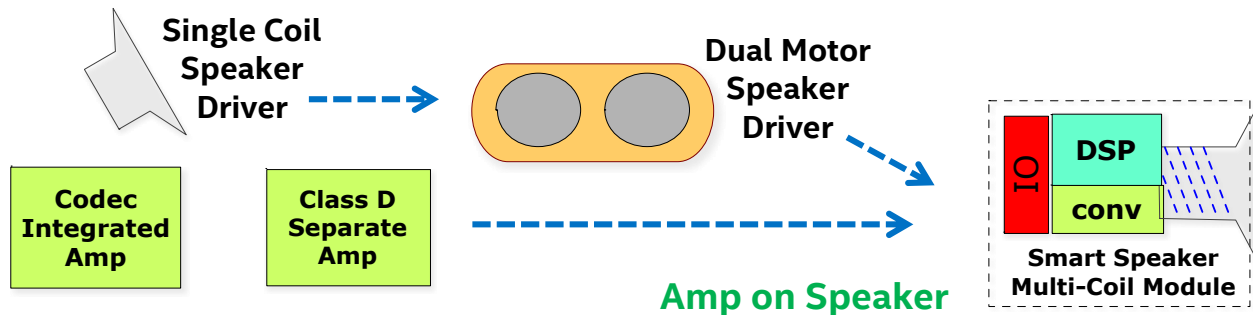
Sensing

Controllable Gain

Arrays

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# Transducers Technology Roadmap

Before

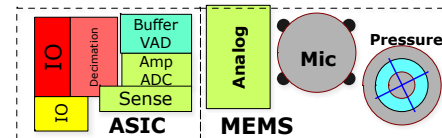
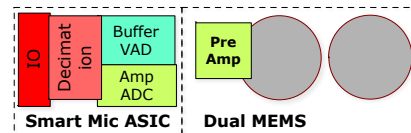
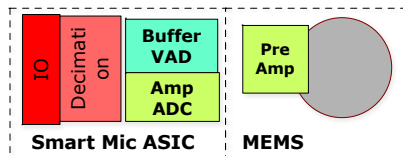
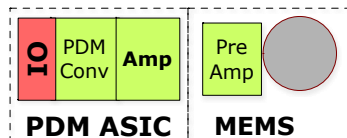
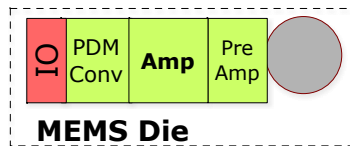
Now

Analog

PDM

I2S

SoundWire\*



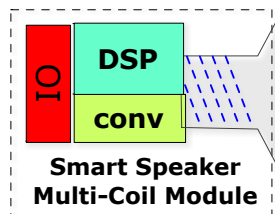
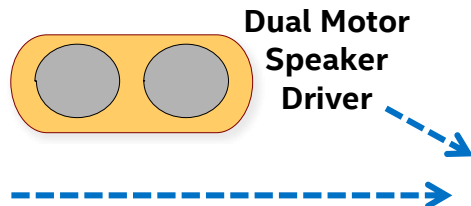
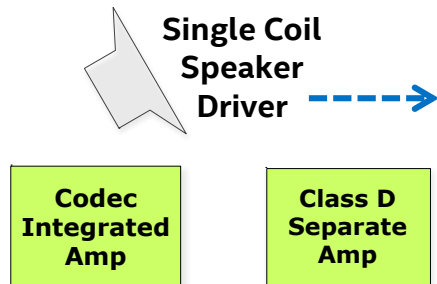
Sensing

Controllable Gain

Arrays

Improved Mounting

Ultrasonics



Amp on Speaker

# Transducers Technology Roadmap

Before

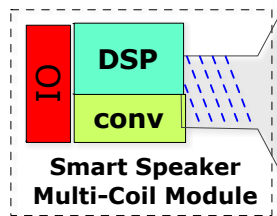
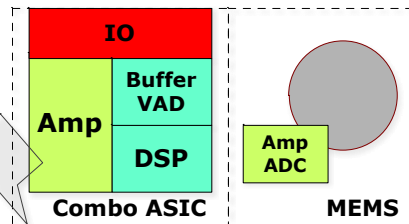
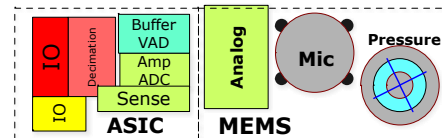
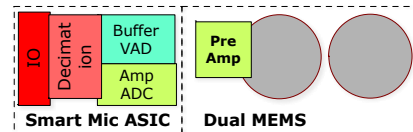
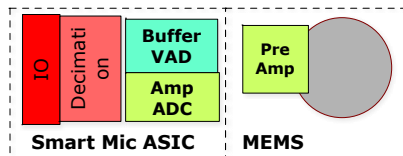
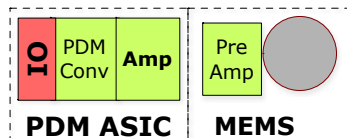
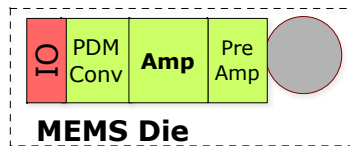
Now

Analog

PDM

I2S

SoundWire\*



Sensing  
Controllable Gain

Ultrasonics

Arrays

Improved Mounting

Amp on Speaker



# Digital Thiele-Small Correction

- **DTSC**, Trigence's\* Digital Thiele-Small Correction technology offers new type of equalizer tool to improve / compensate the electro-mechanical characteristics of speakers through the precise real-time measurement of the speaker properties, and use of on-chip DSP
- Uses physical model of transducer and modern feed-forward control theory to permit the time-domain equalization of speaker response
- Easy to produce preferred sound flavor
  - Stronger or slower bass
  - Fast or slow transient attack
  - Simple boost up or down of mid-high
  - Short or long reverberation
  - Clarity or warm sound

