



# Microphone Beamforming

Syntronic Presentation For Chalmers DAT096

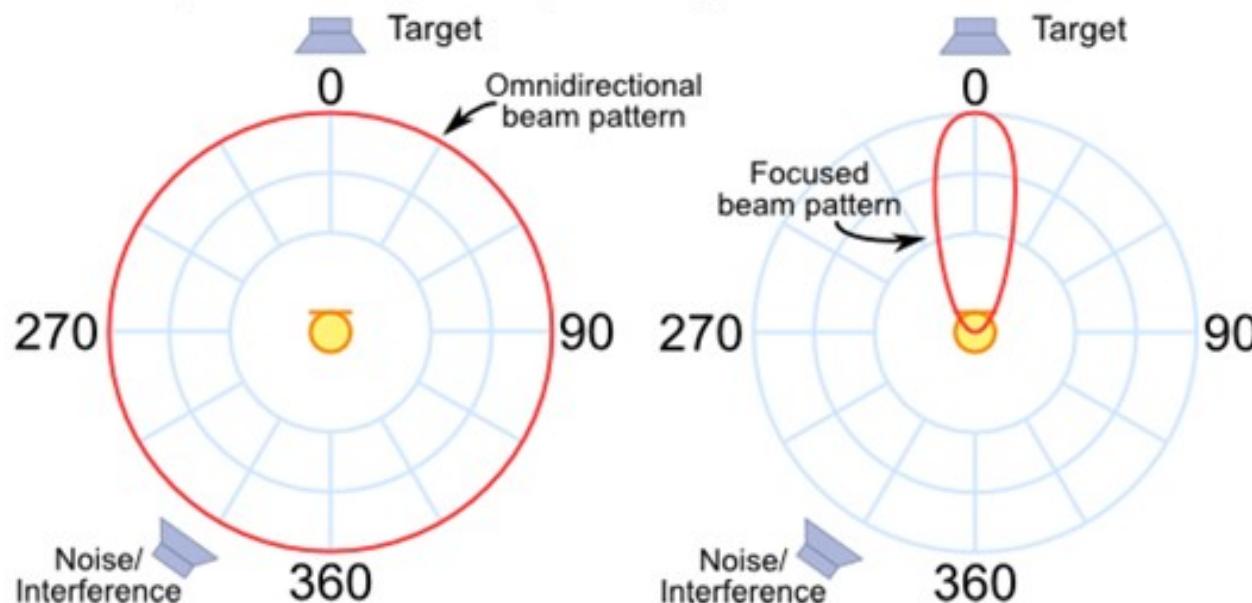
Steinar Þorvaldsson  
Bhavishya Goel



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UNIVERSITY OF TECHNOLOGY

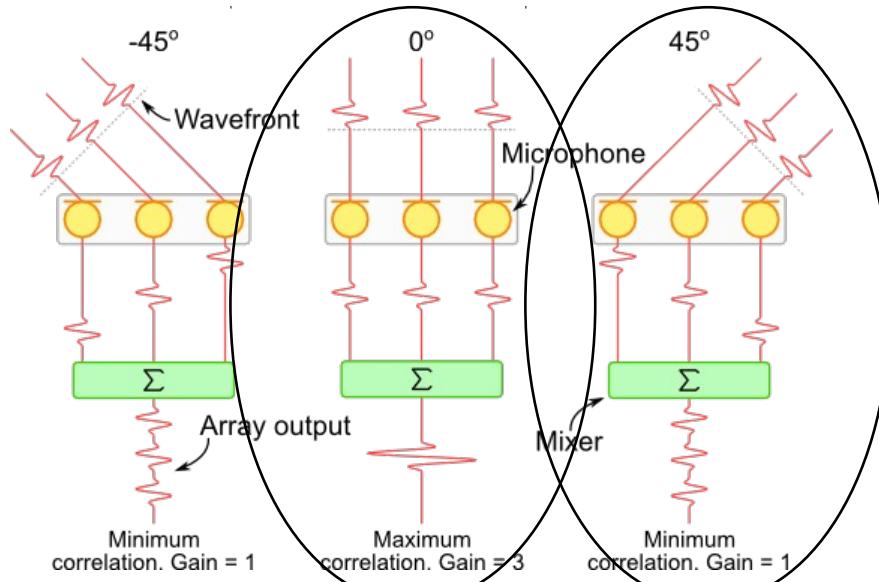
# What is Beamforming?

- A technique to achieve directionality in signal transmission or **reception**



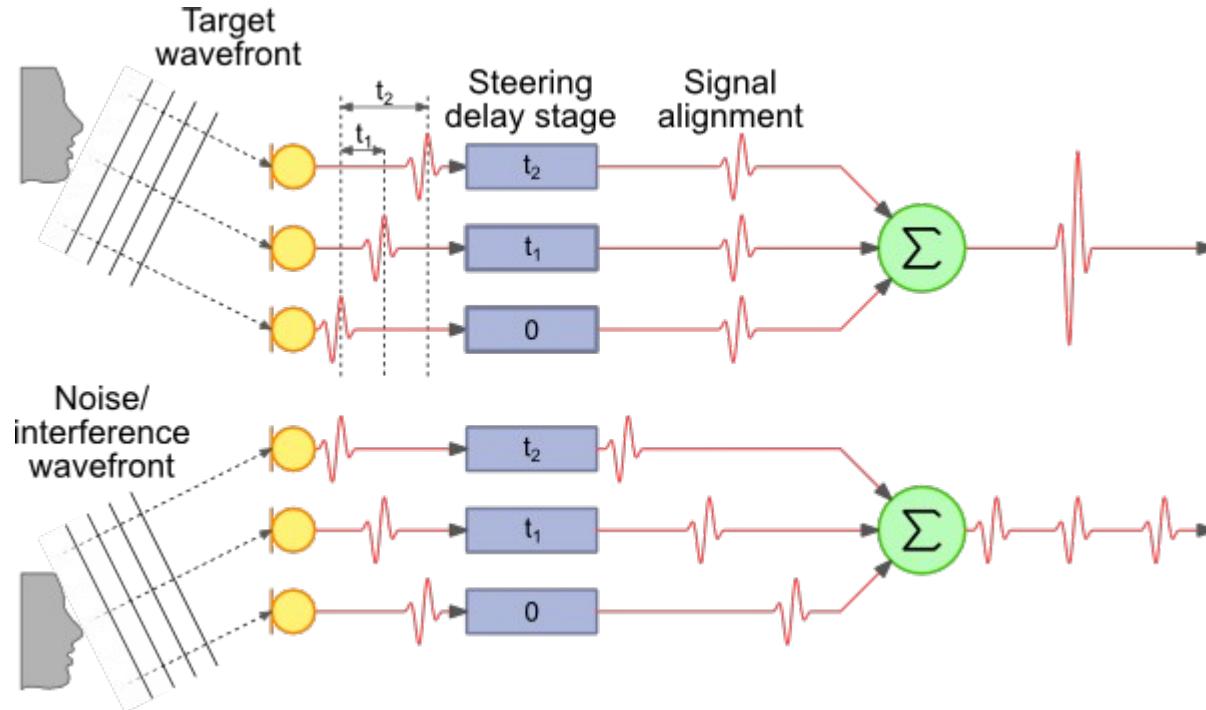
# Beamforming Basics

- Directionality achieved by combining inputs from multiple sensors
- The signal coming from a desired angle is boosted compared to the signals coming from undesired angles

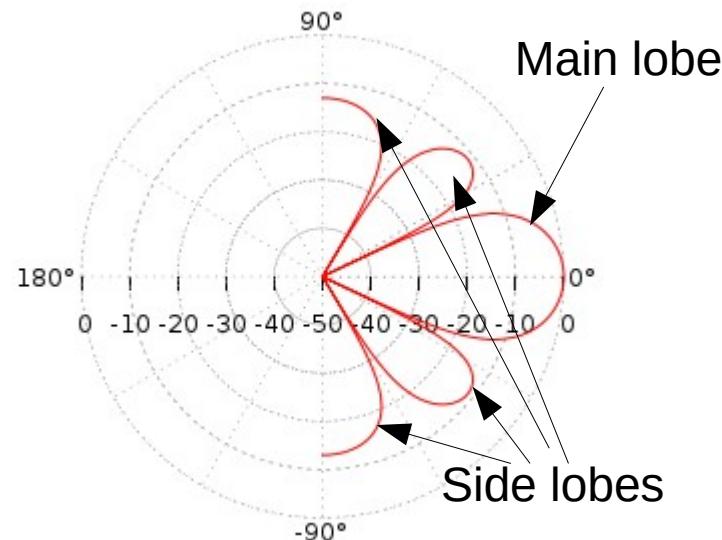
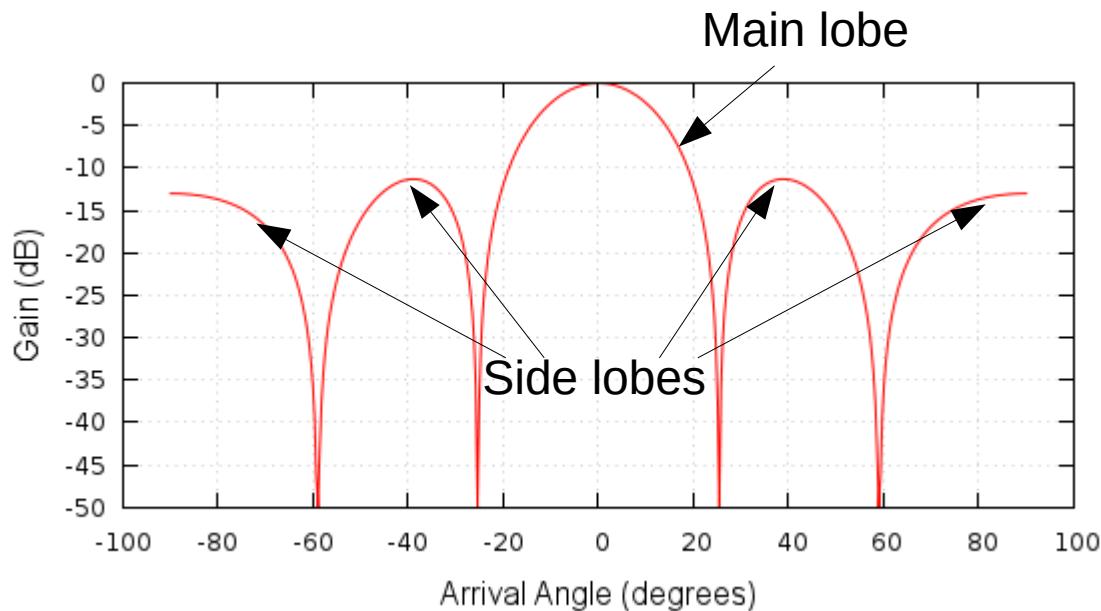


# Beamforming Basics

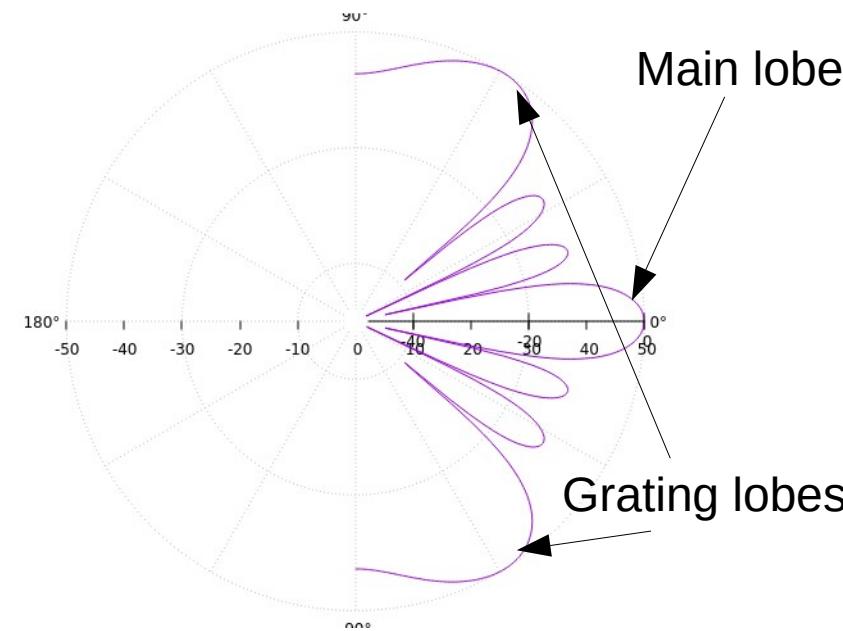
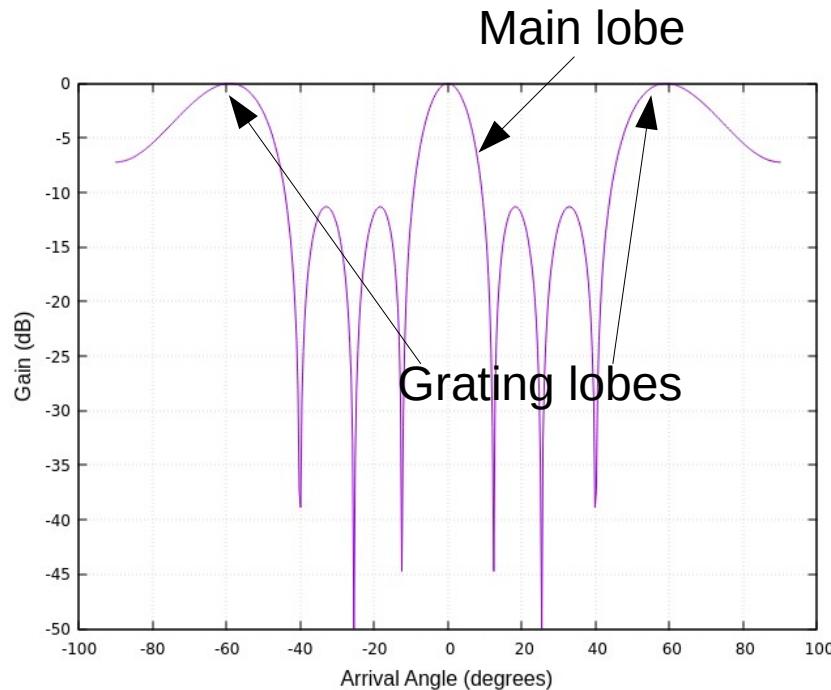
- Simplest beamforming algorithm: Delay and sum



# Delay and Sum Beamformer Beam Pattern



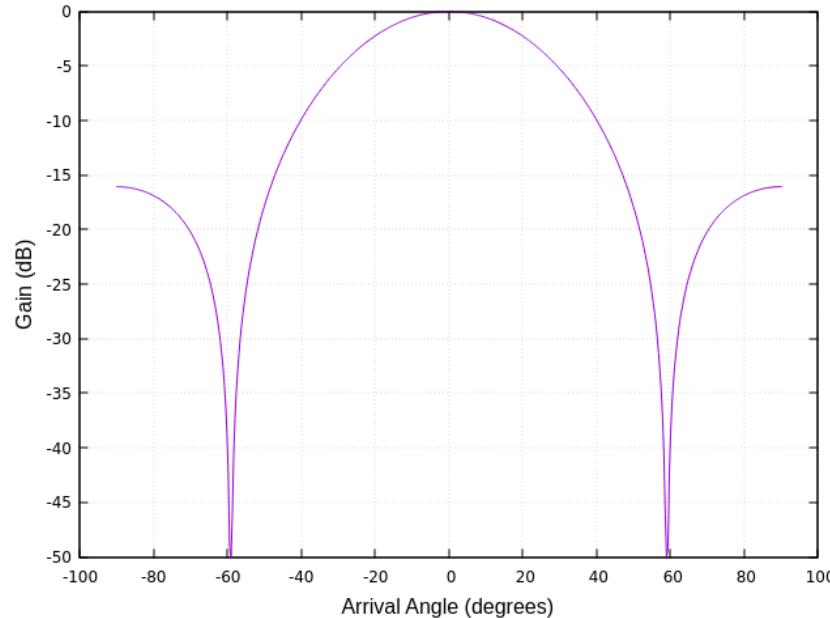
# Delay and Sum Beamformer Beam Pattern



# Delay and Sum Beamformer Characteristics

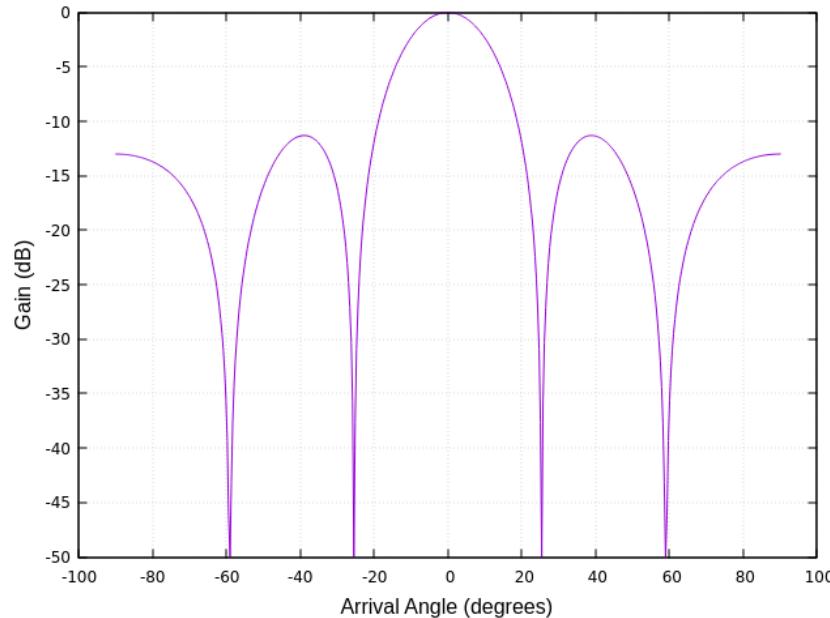
- The beam pattern depends upon following parameters:
  - Speed of sound (c)
  - Number of microphones (n)
  - Distance between microphones (d)
  - Frequency of the incoming signal (f)
- Spatial aliasing
  - Results in grating lobes
  - $f_{\max} = c/d$
  - $d_{\max} = c/f$

# Effect of Distance between Microphones



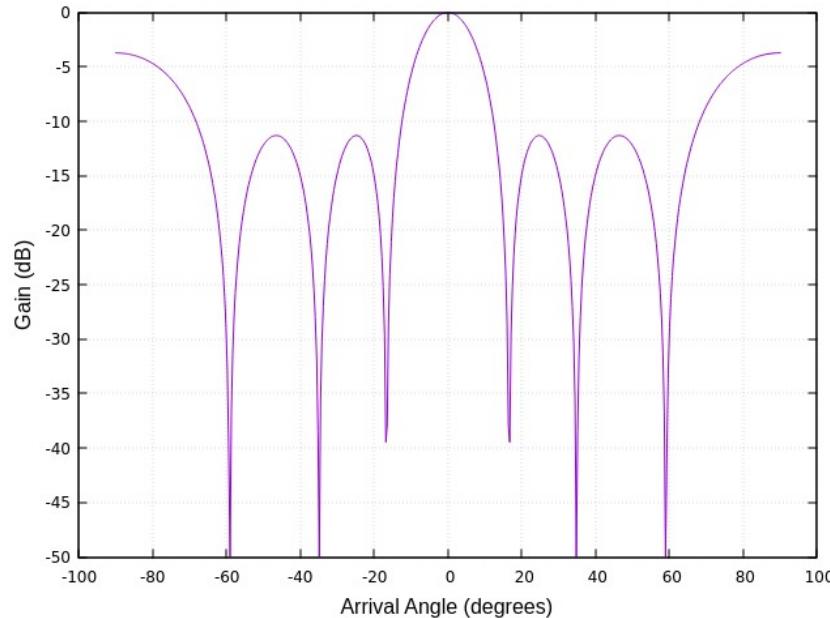
$$d = 100\text{mm}, n = 4, f = 1\text{kHz}$$

# Effect of Distance between Microphones



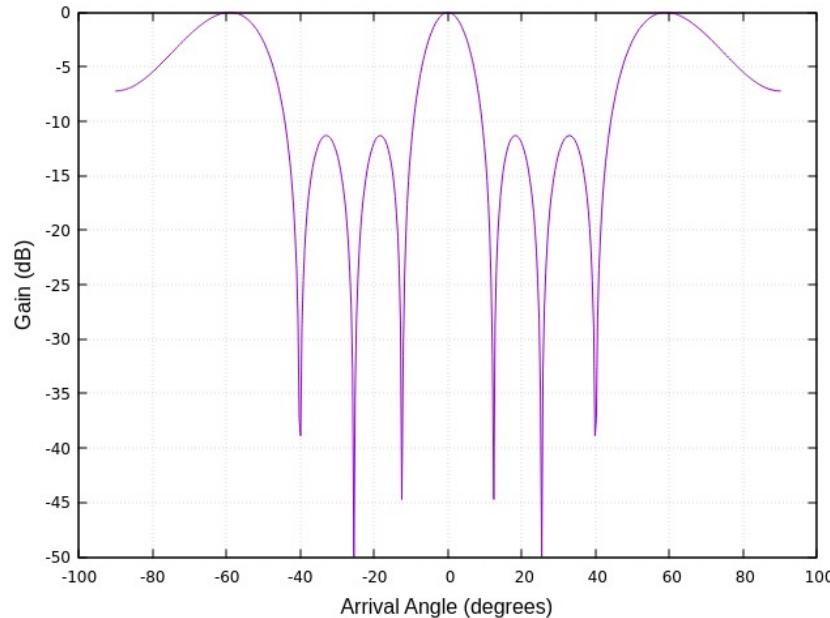
$$d = 200\text{mm}, n = 4, f = 1\text{kHz}$$

# Effect of Distance between Microphones



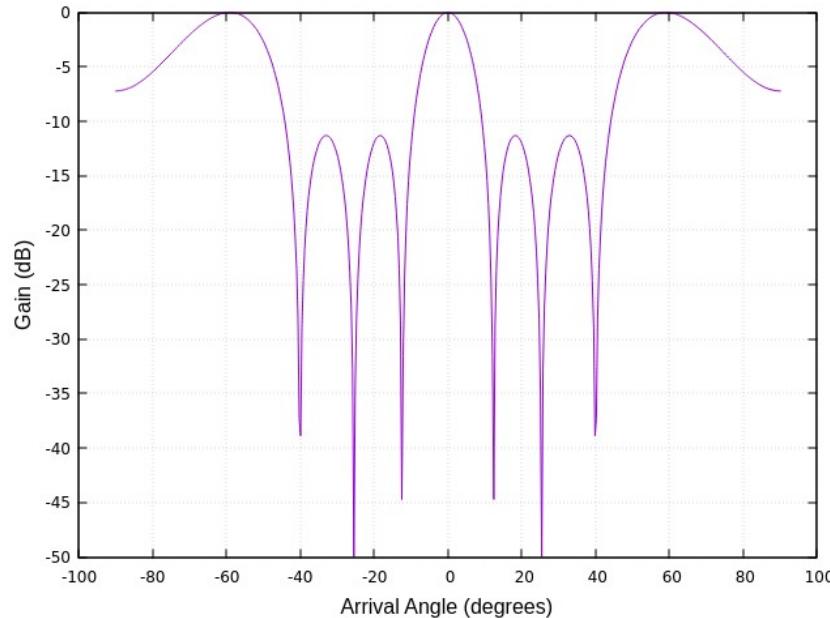
$$d = 300\text{mm}, n = 4, f = 1\text{kHz}$$

# Effect of Distance between Microphones



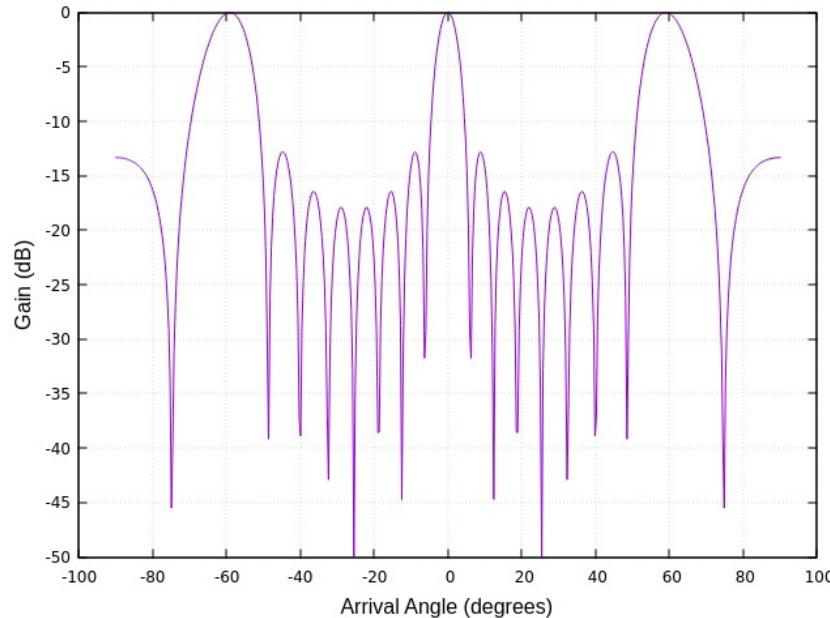
$$d = 400\text{mm}, n = 4, f = 1\text{kHz}$$

# Effect of Number of Microphones



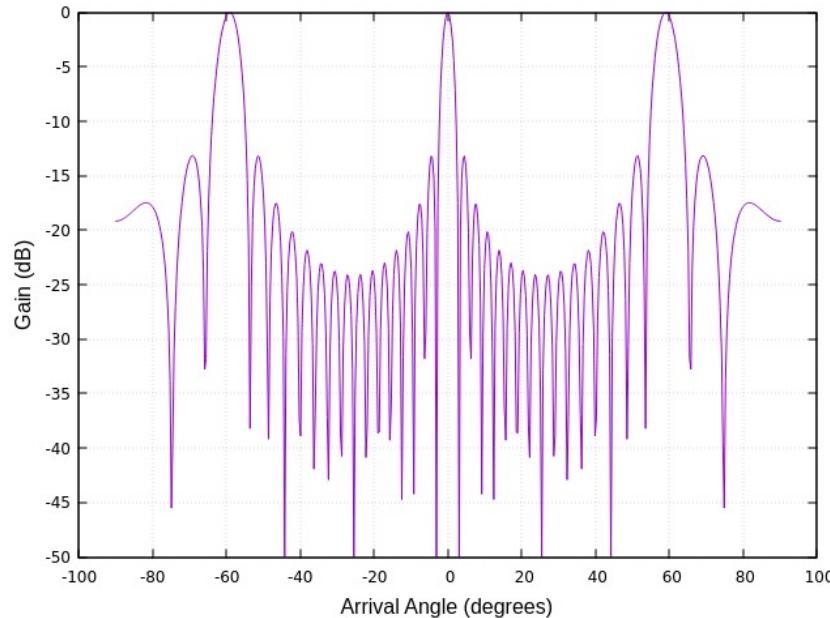
$$d = 400\text{mm}, n = 4, f = 1\text{kHz}$$

# Effect of Number of Microphones



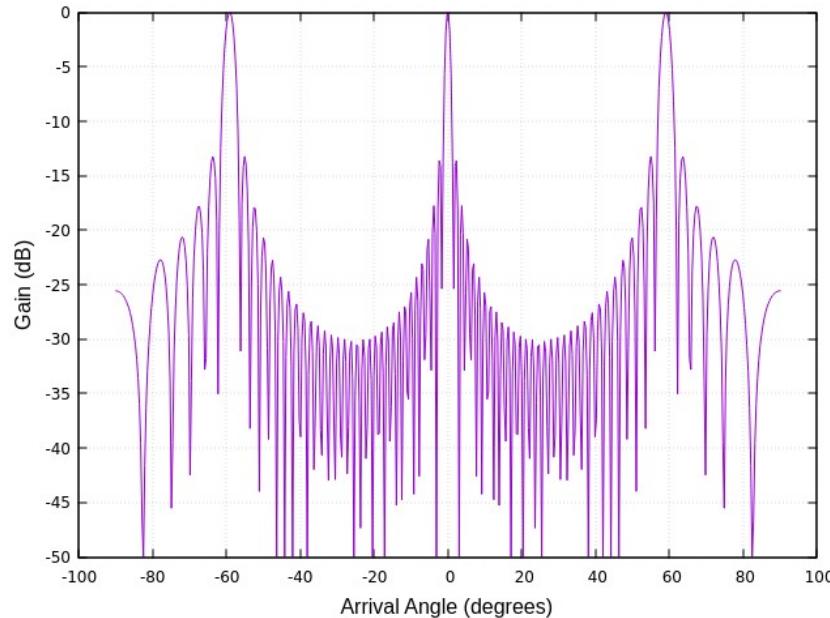
$d = 400\text{mm}$ ,  $n = 8$ ,  $f = 1\text{kHz}$

# Effect of Number of Microphones



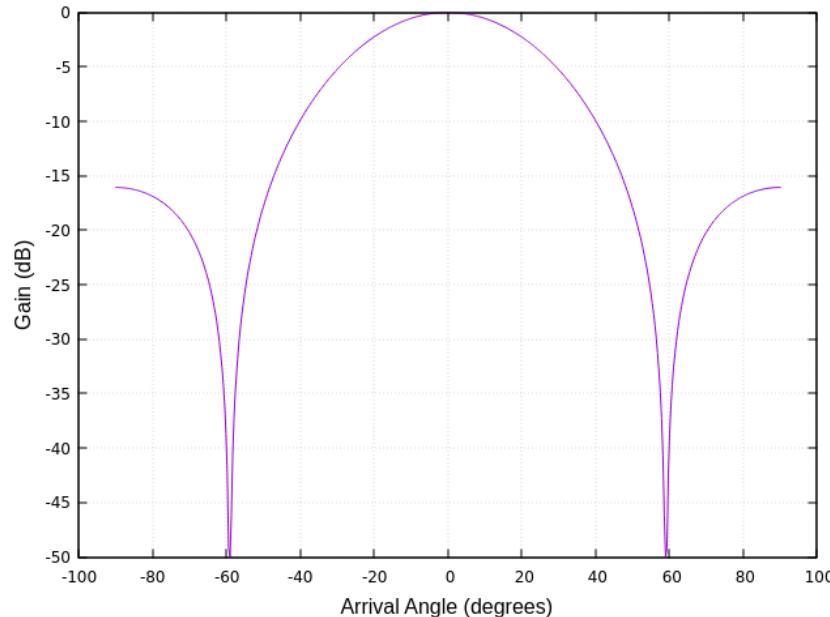
$d = 400\text{mm}$ ,  $n = 16$ ,  $f = 1\text{kHz}$

# Effect of Number of Microphones



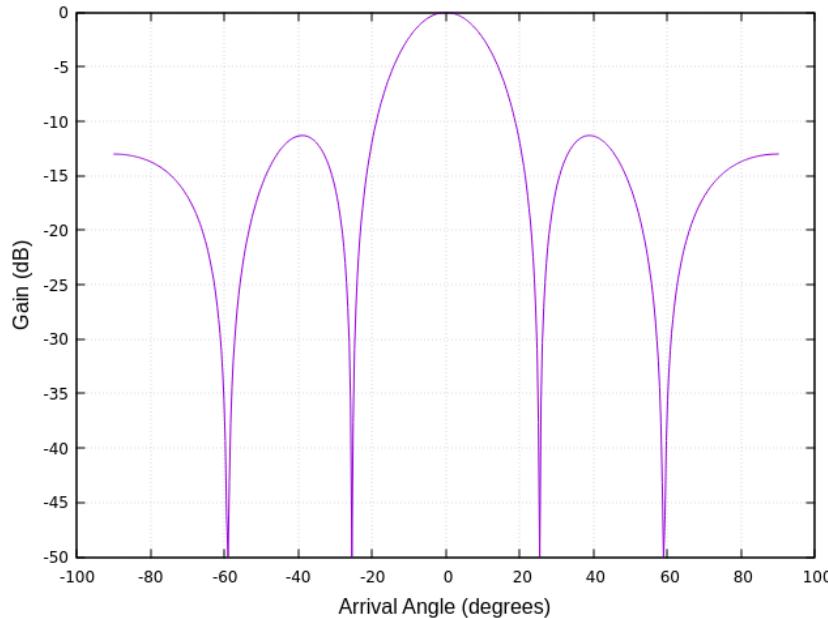
$d = 400\text{mm}$ ,  $n = 32$ ,  $f = 1\text{kHz}$

# Effect of Input Frequency



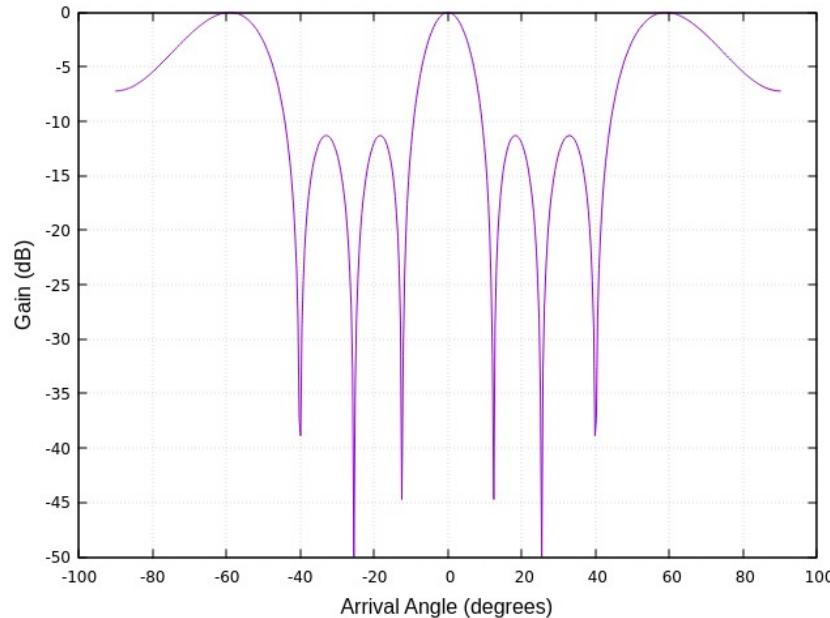
$$d = 100\text{mm}, n = 4, f = 1\text{kHz}$$

# Effect of Input Frequency



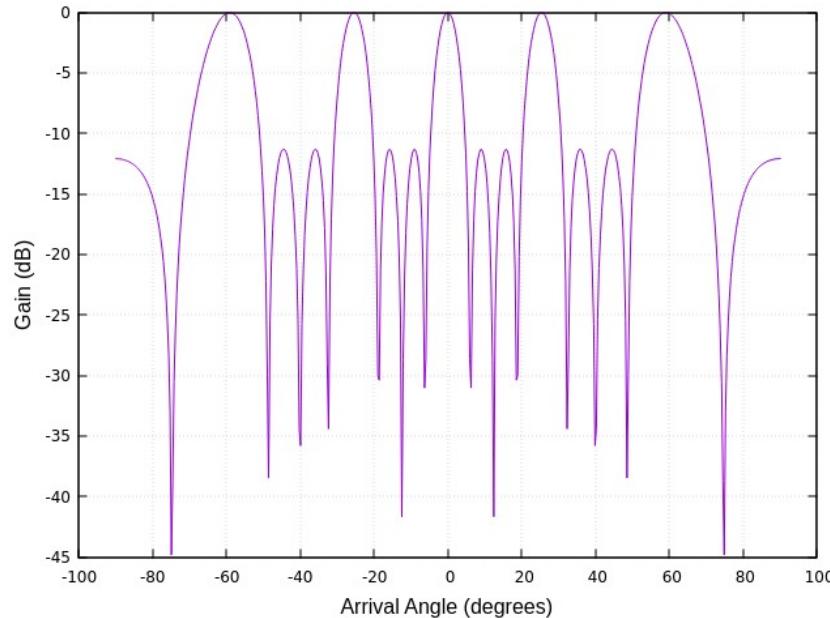
$$d = 100\text{mm}, n = 4, f = 2\text{kHz}$$

# Effect of Input Frequency



$$d = 100\text{mm}, n = 4, f = 4\text{kHz}$$

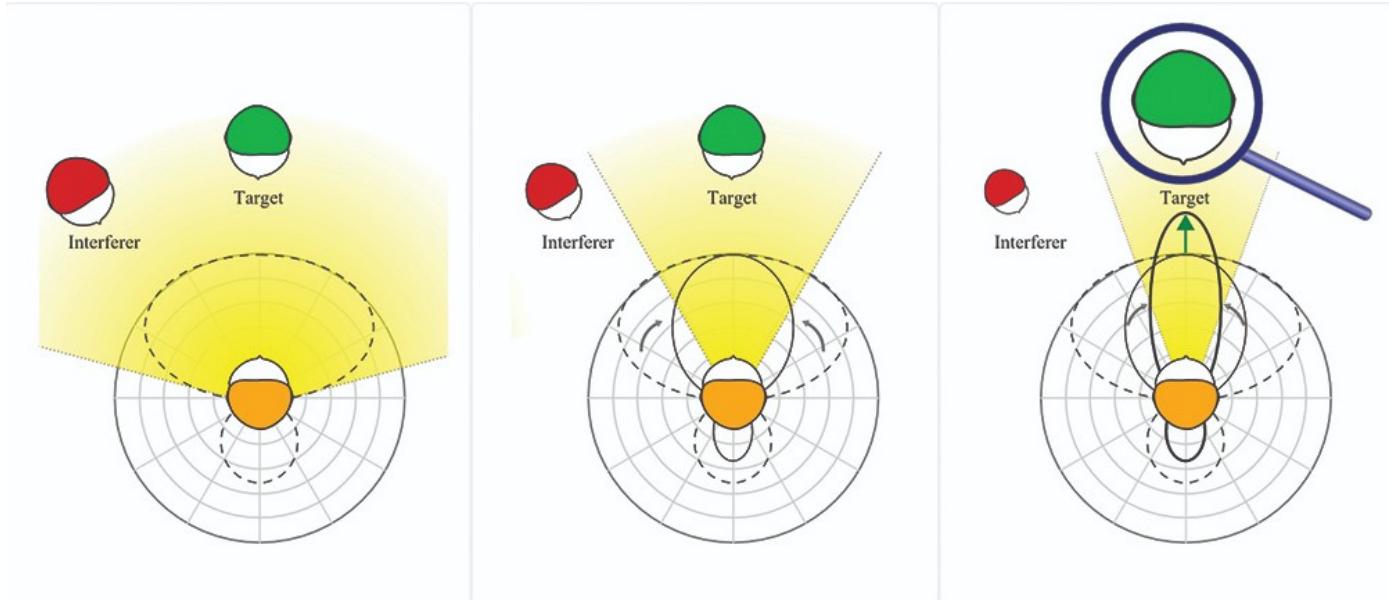
# Effect of Input Frequency



$$d = 100\text{mm}, n = 4, f = 8\text{kHz}$$

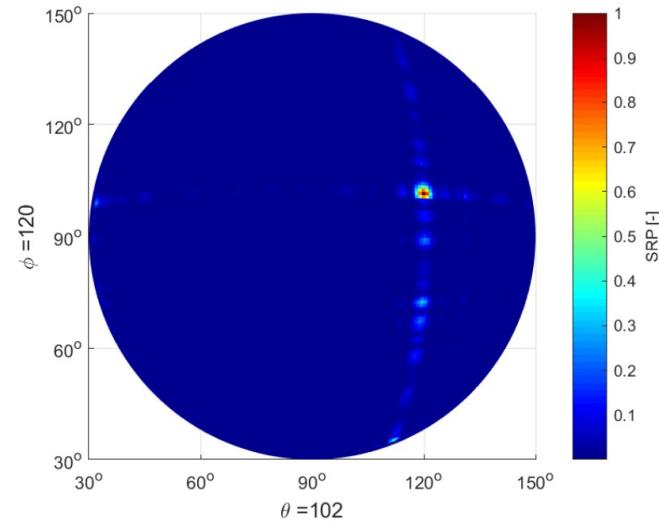
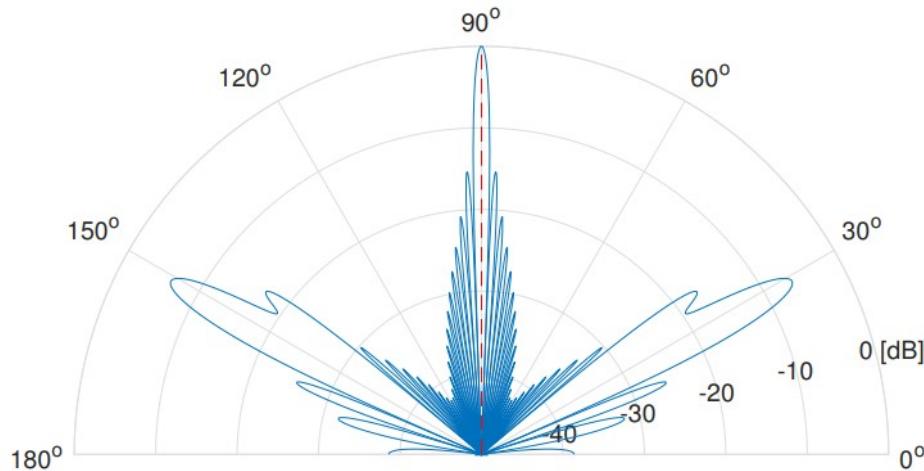
# Applications

- Directional Speech enhancement



# Applications

- Sound source localization (Direction of Arrival)



# Arrays

- Represent how you sample signals “in space”. Can be in 1, 2 or 3 dimensions and consist of 2 or more sensors
- Shape of an array will affect how to solve for the directionality of the array.
- The set of delays which maximally add together the wavefront of an incoming wave at a certain angle is the current “viewing” direction of the array.
- Distance between microphones will dictate what frequencies the array can be optimized for.

# Beamforming Algorithms - Adaptive

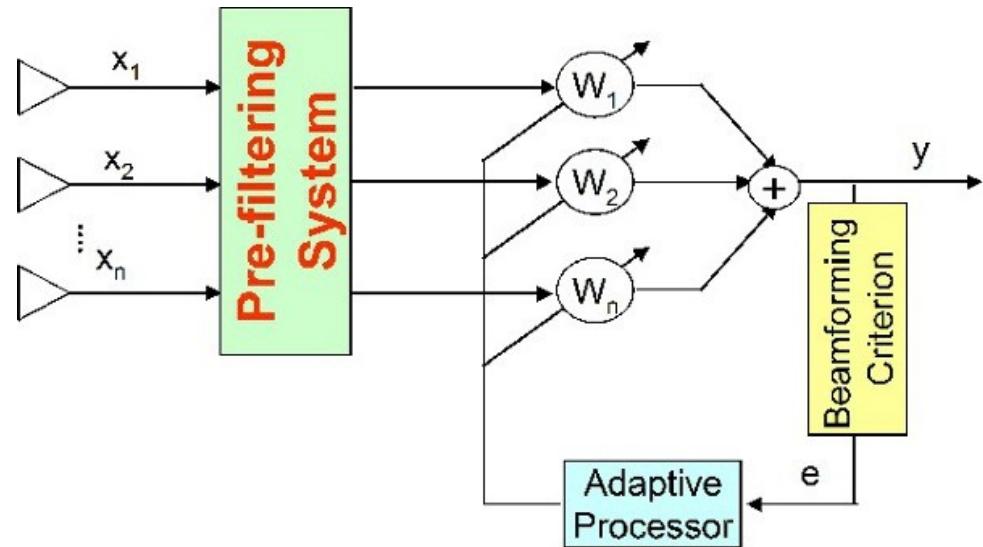
- Adaptively modify the delay and weights of the beam to achieve optimal result
- Adjust the beam to cancel interference in comparison to target, ie nulling.
- Generally more complex

Examples:

Least-Mean-Squares

Recursive-Mean-Square

Generalised SideLobe Canceler



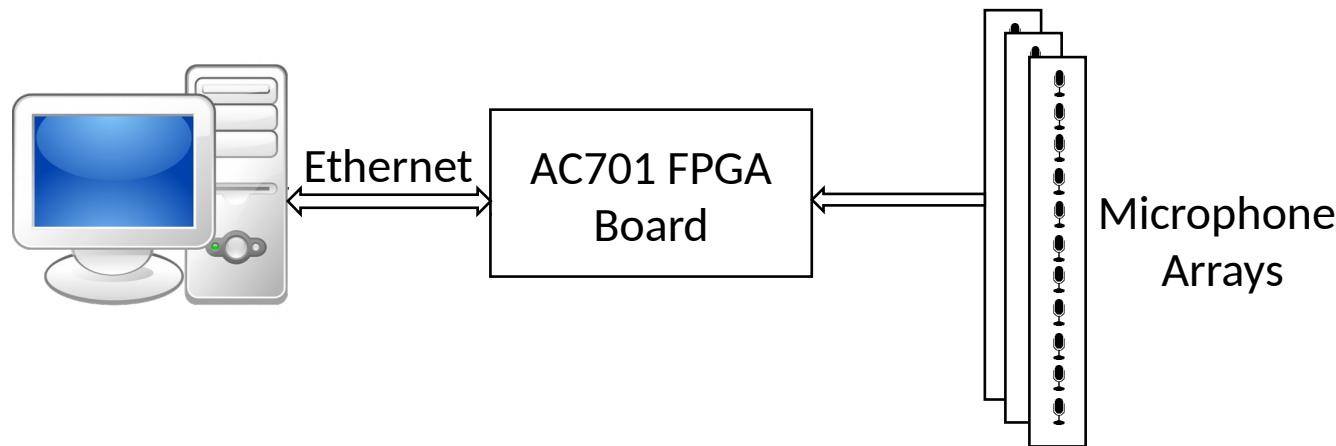
# Algorithm differences between applications

- Speech enhancement and isolation involves more around selecting weights/delays to constructively and destructively add waveforms together to either enhance certain sources over others or separate sources.
- Localization involves more around scanning and searching. With focus on finding the beam orientation with maximum input signal power

# Real time suitability of algorithms

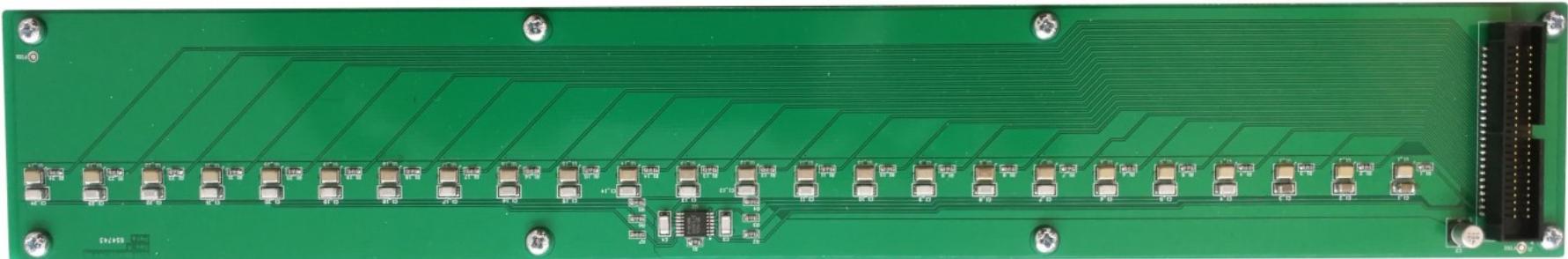
- Not all algorithms are suitable for real time systems.
- Complex mathematics difficult to implement on FPGA
- Cross correlation algorithms grow exponentially with number of sensors in array (grid search localization).
- FPGAs have limited resources.

# Hardware Design for DAT096



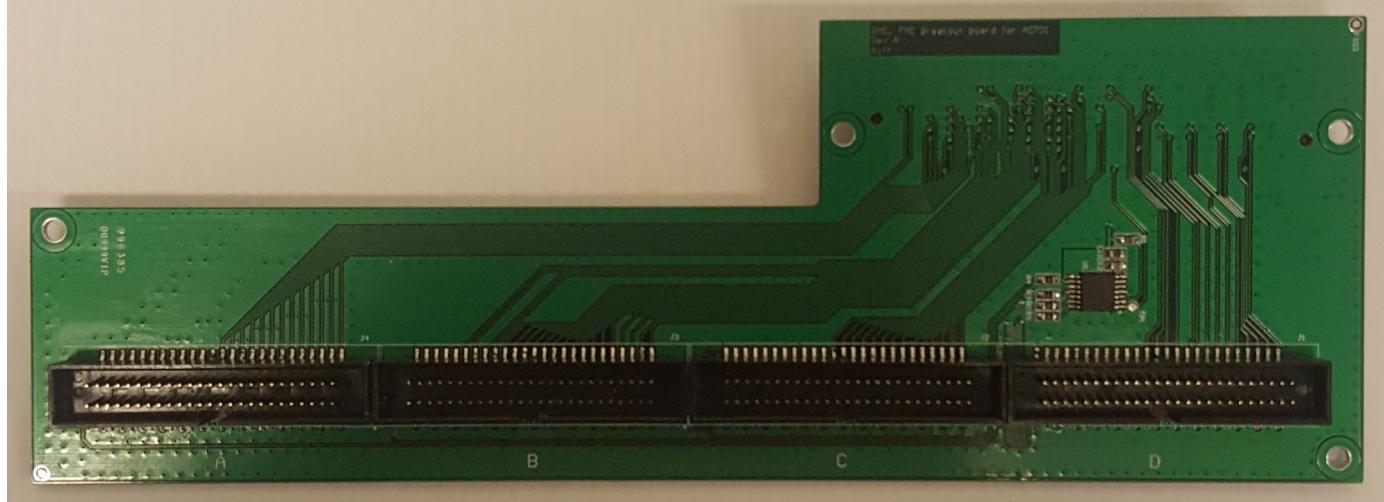
# Syntronic Hardware - Linear Arrays

- Linear array of 24 MEMS microphones.
- Good for testing and development, but may not be optimal for speech.
- D=11.43mm
- You can make your own! We have schematics and guidelines on how to best achieve this.



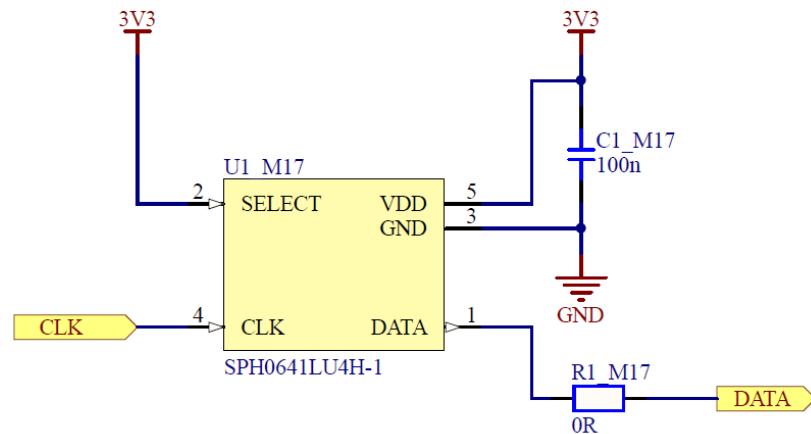
# Syntronic Hardware - FMC Breakout

- FMC Breakout board for AC701
- Support for up to 4 separate panels with 24 microphones each, or one with 96
- Breaks out FPGA signals on FMC connector and propagates clock to all MEMS microphones



# Syntronic Hardware - Microphone

- The Syntronic Array uses an Ultrasonic microphone SPH0641LU4H-1.
- Can be clocked at different frequencies to get different bandwidth. 20Hz to ~80Khz @ 4.8Mhz
- Oversampled output in PDM format,  
Needs to be filtered to get PCM format



Microphone Channel

Thank you for having us!

Questions ?

