



# Microphone Beamforming

Syntronic Presentation For Chalmers DAT096

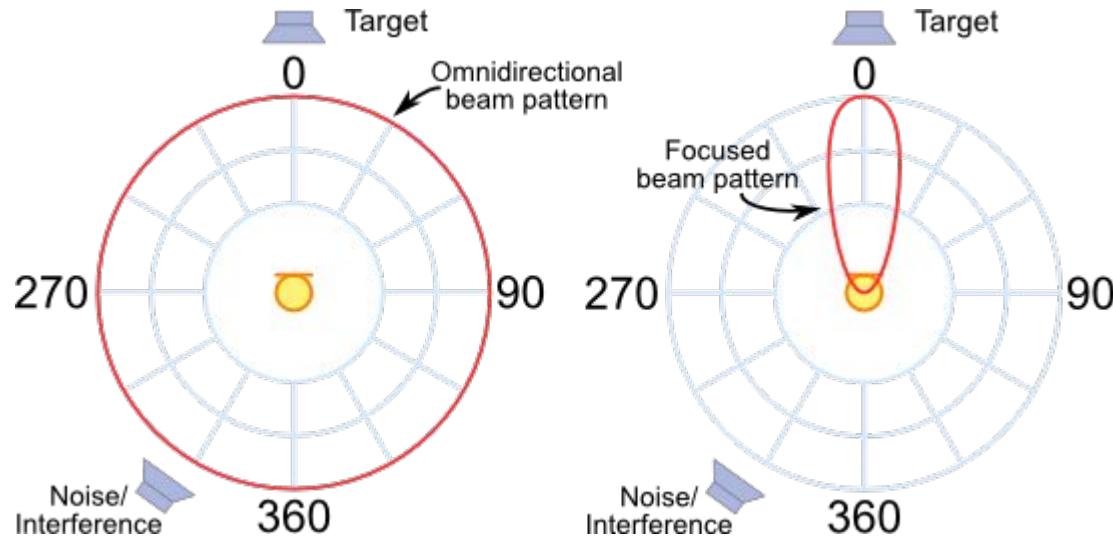
Steinar Þorvaldsson  
Bhavishya Goel



**CHALMERS**  
UNIVERSITY OF TECHNOLOGY

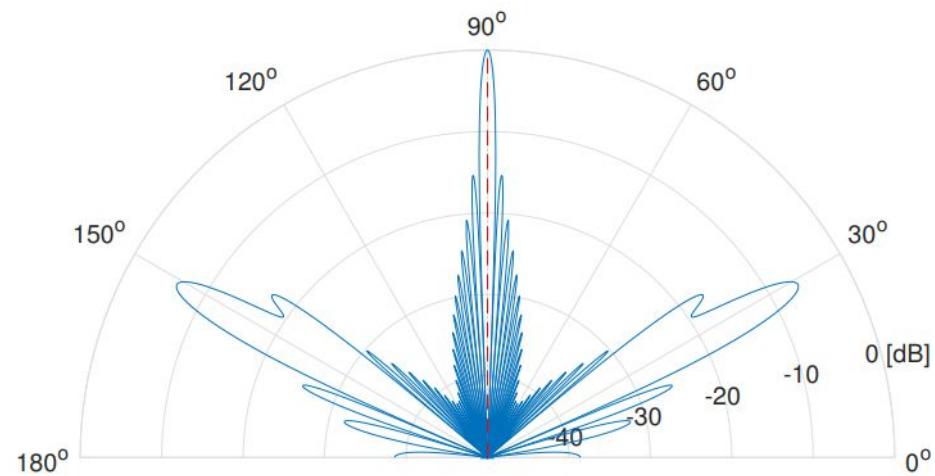
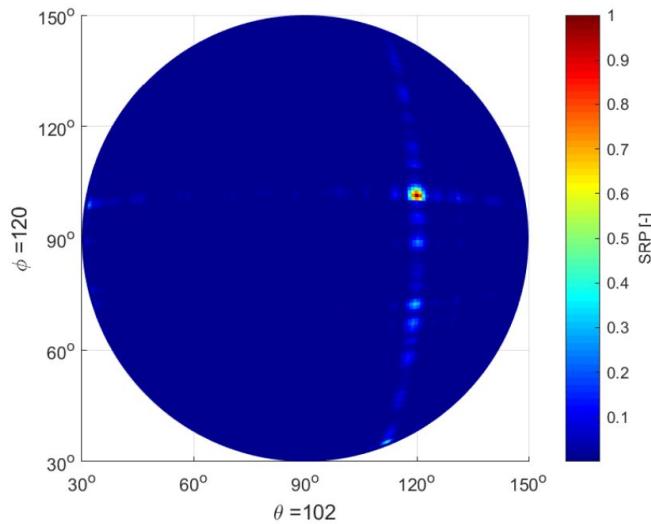
# What is beamforming ?

-Sampling information in time and space and applying signal transformations in order to achieve spatial filtering.



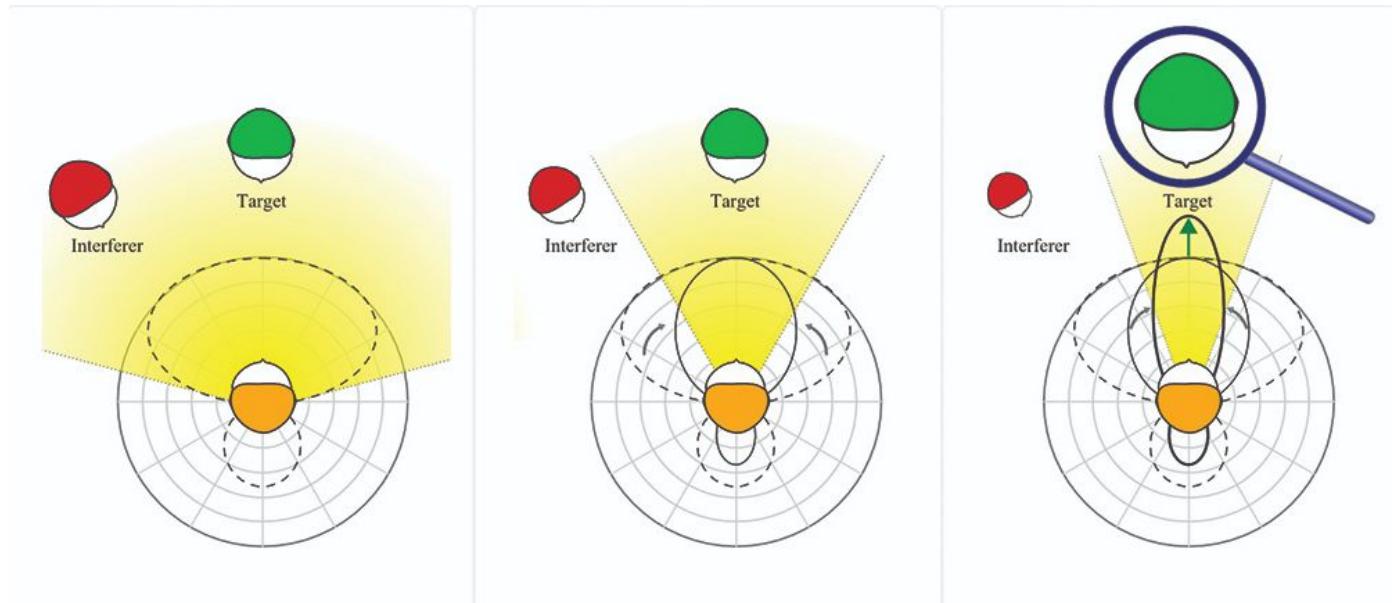
# Applications

-Source sound localization (Direction of Arrival)



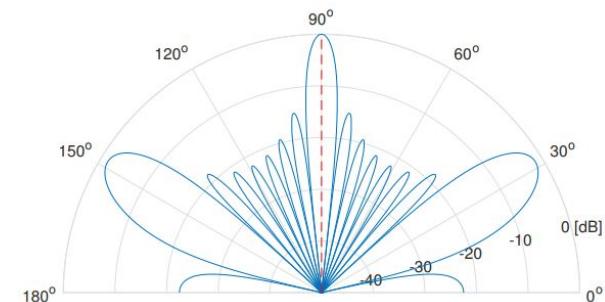
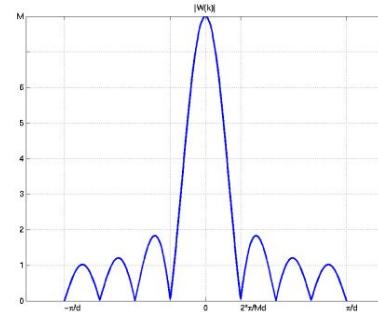
# Applications

-Directional Speech enhancement



# Nuances of spatial sampling

- Spatial sampling theorem similar to nyquist applies to spatially sampled signals in order to prevent aliasing.
- Signals must be sampled in space with distance  $d < \lambda/2$
- More important for narrowband signals than wideband signals



# 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 - Conventional

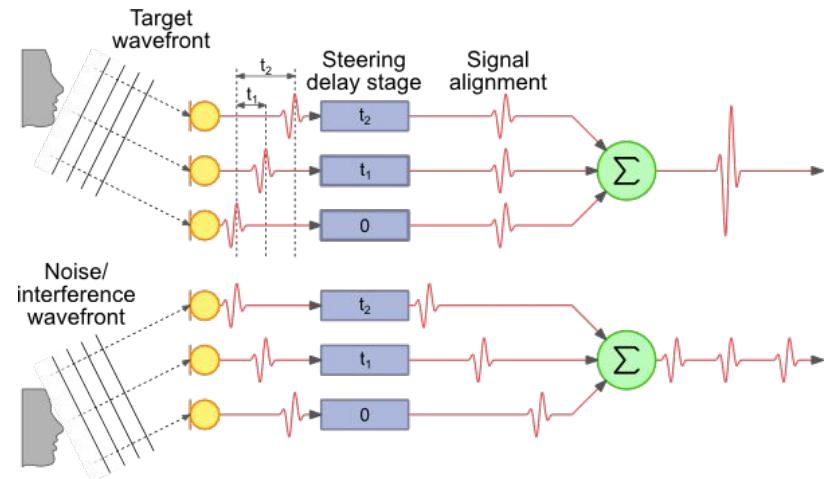
-Fixed weighing and processing of signals to produce spatial filtering effect.

Examples:

Delay and Sum

Filter and Sum

Frequency Domain Beamforming



# Beamforming Algorithms - Adaptive

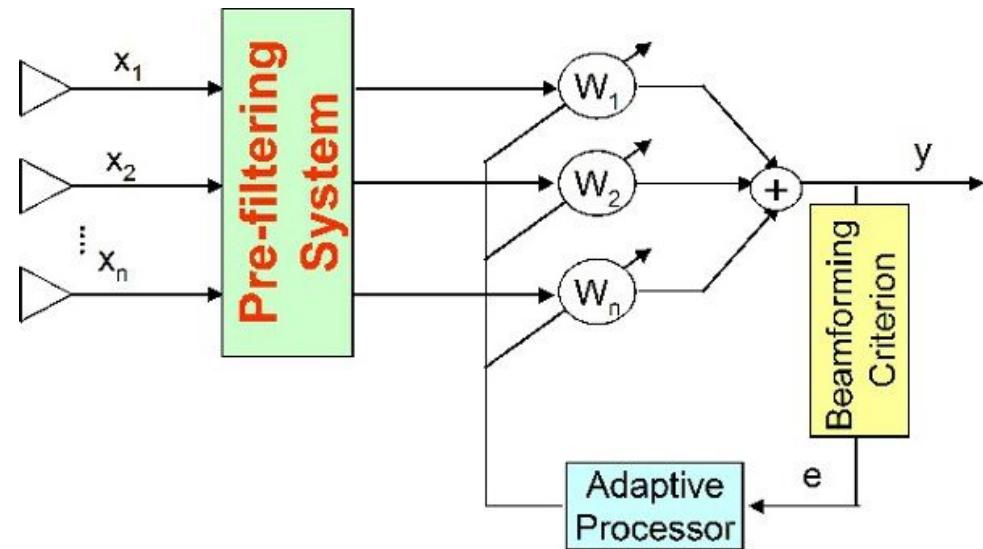
- Adaptively modify the delay and weights of the beam to achieve optimal result
- Adjust the beam to cancel interference in comparison to target, ie nulling.
- Generally more complex

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

Example Source-Localisation Algorithms:  
SRP-PHAT

# 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.

# System Dynamics

Designing the system is a tradeoff between requirements, performance and resource use.

## **-Increasing sampling speed**

Causes Increased resource usage [Filter lengths]

Decides minimum difference in delay possible between two signals (beam resolution).

Allows higher system SNR with decimation and filtering

## **-Increasing amount of sensors**

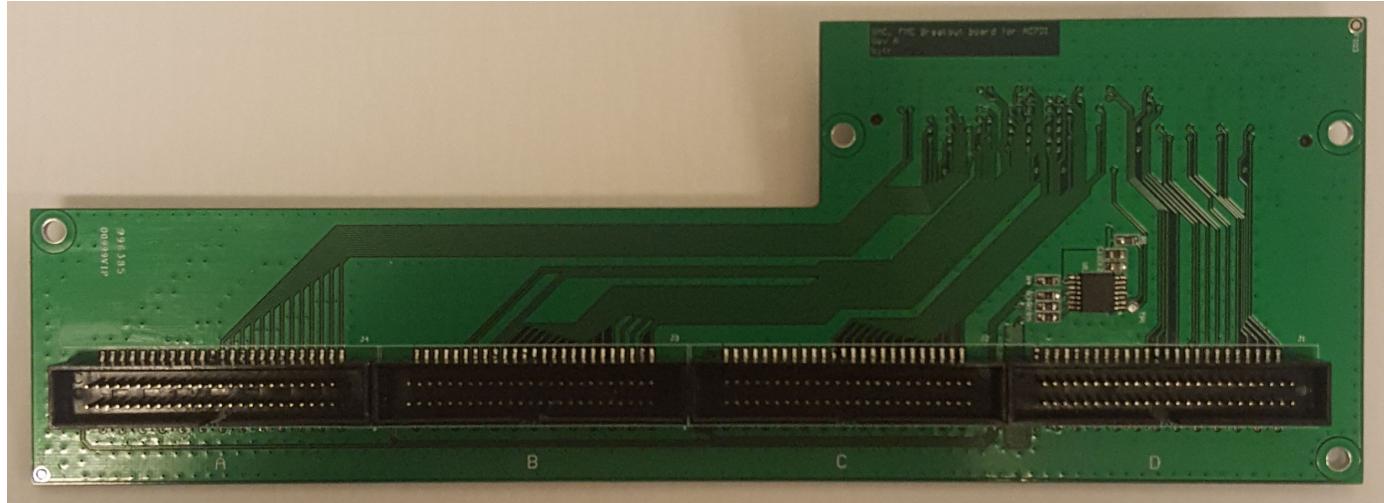
Causes increased resources usage [Amount of filters]

Allows narrower beams

Allows higher system SNR

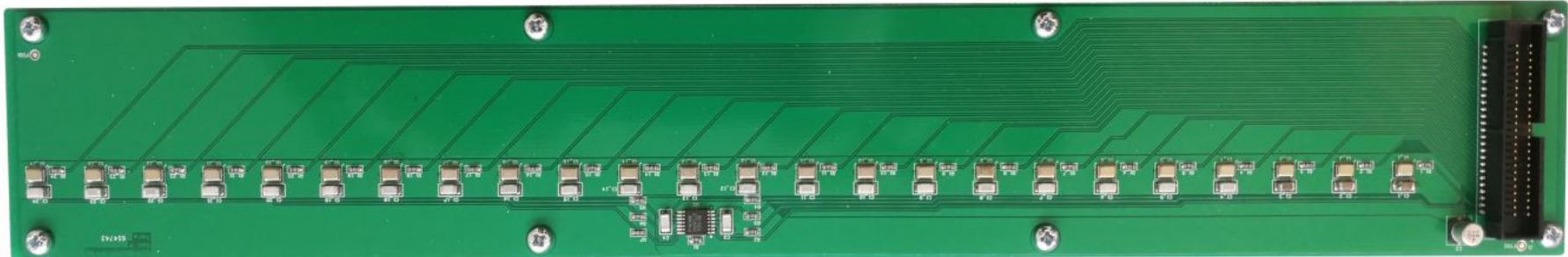
# Syntronic Hardware - FMC Breakout

- 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 - Linear Arrays

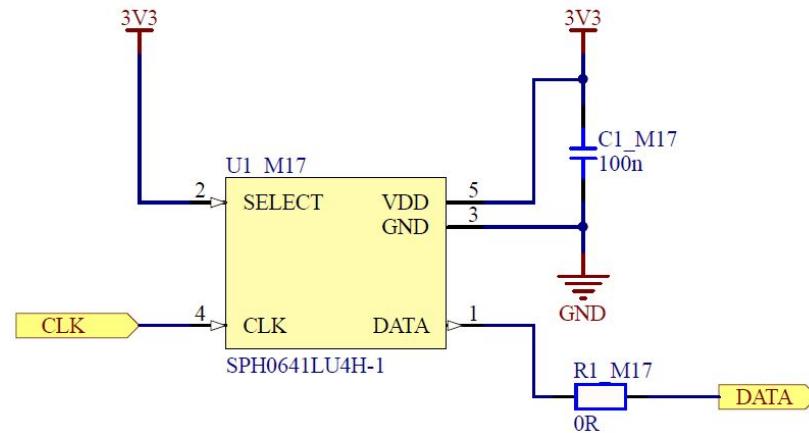
- Linear array of 24 MEMS microphones.
- Good for testing and development, but may not be optimal for speech.
- D=11.43mm
- You can make your own! We have schematics and guidelines on how to best achieve this.



# Syntronic Hardware - 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 ?

