

A digitally controlled two dimensional loudspeaker array

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The full staff: LAE and..

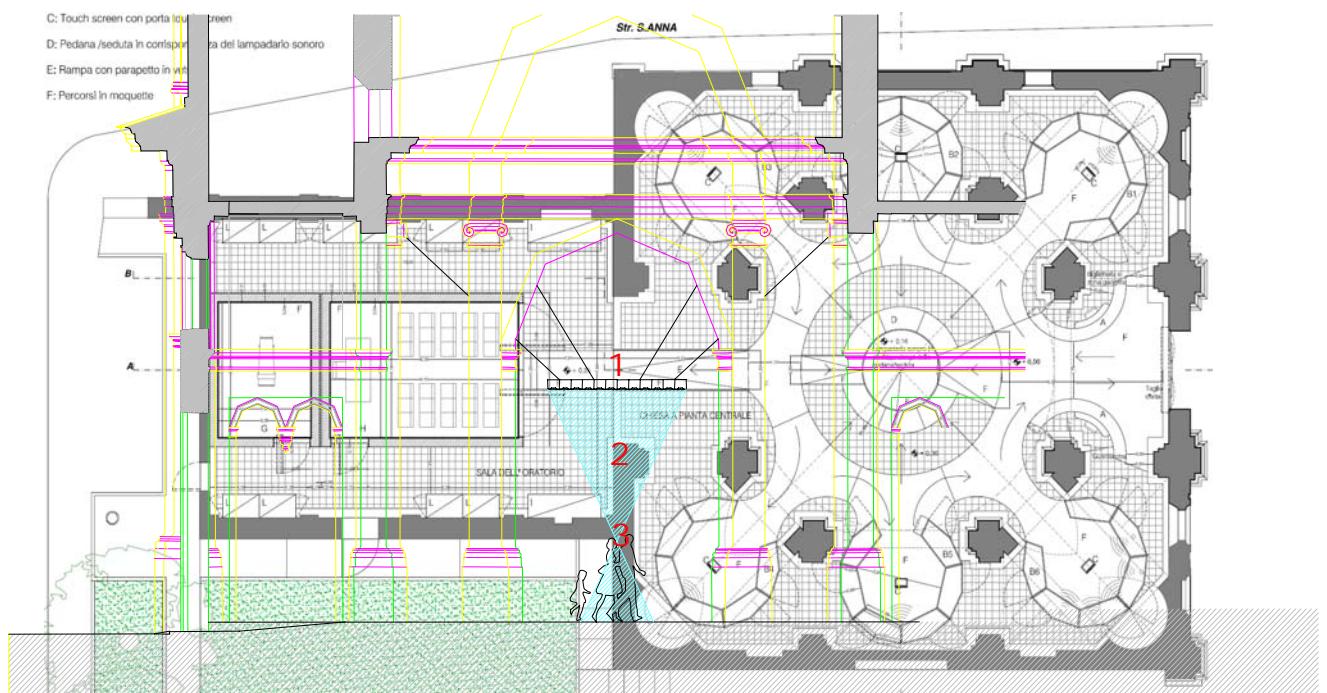
- **Paolo Martignon** Project management, geometry and filter design
- **Daniele Torelli** SW development on Linux PC
- **Fons Adriaensen** Consultance on SW, HW and algorithms
- **Roberto Zana** Wiring design and realization
- **Audiolink srl and AIDA srl, Parma** Place, instrumentation, hardware help and consultance
- **Aldo Piazza** Chandelier iron structure realization

The project: a "sonic chandelier"



- S. Elisabetta church (reverb time 5 sec)

- Sonic chandelier:



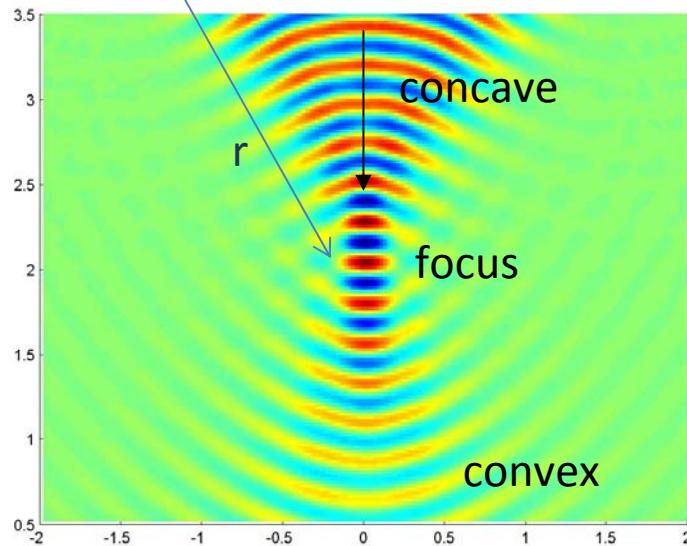
Presentation outline

- Wave Field Synthesis concepts, spatial aliasing
- Speakers choice and line array prototype
- 2D array design and realization
- HW description and signal processing scheme
- Filters structure, design and implementation
- Validation test and measurement

Sound focalization by WFS

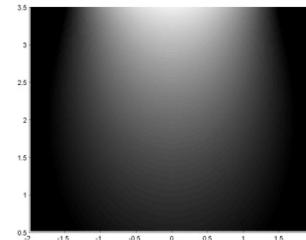
(Delft University of Technology, 90's)

Array

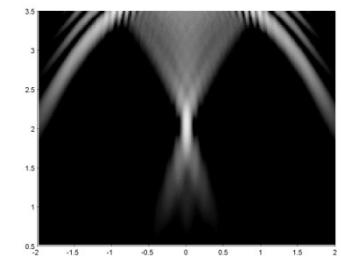


Sensible parameters:

$$\frac{A}{\lambda} \quad \text{Set the beam width..}$$



$$\frac{\Delta x}{\lambda} \quad \text{Spatial aliasing, secondary lobes..}$$



The front curvature is obtained by means of a gain-delay set ..

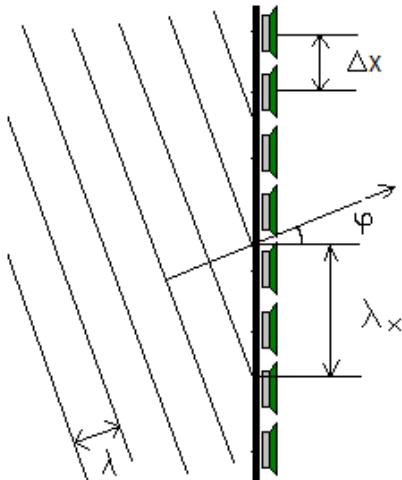
$$Q_m^{foc}(x, \omega) = S(\omega) \sqrt{\frac{k}{2\pi i}} \sqrt{\frac{\Delta z_0}{\Delta z_0 - z_0}} e^{+ikr} \cos \varphi \frac{1}{\sqrt{r}}$$

Common filter

Delays →

Gains →

The spatial aliasing problem



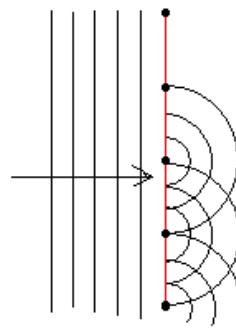
The spatial Nyquist theorem:

Sampling condition: $\lambda_x = \frac{\lambda}{\sin\varphi} > 2 \cdot \Delta x$

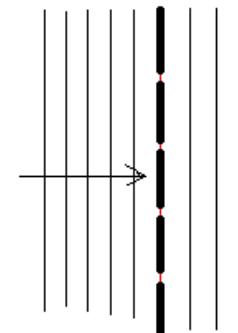
antialiasing sampling filter $\Rightarrow f_{\max} \leq \frac{c}{2\Delta x \sin\varphi}$

Reconstruction condition:

$$\sin \vartheta_{em}(f) \leq \frac{c}{2\Delta x f}$$

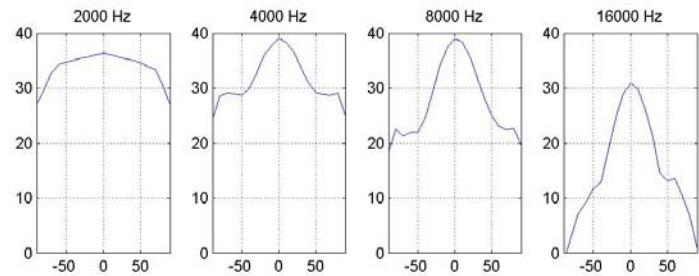
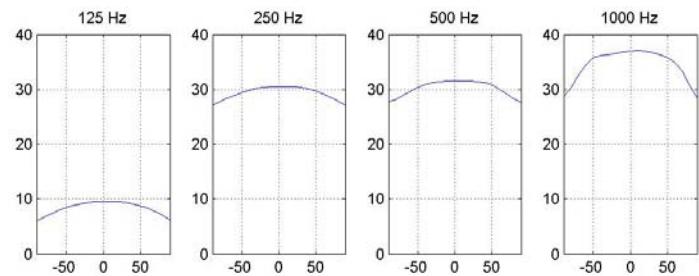
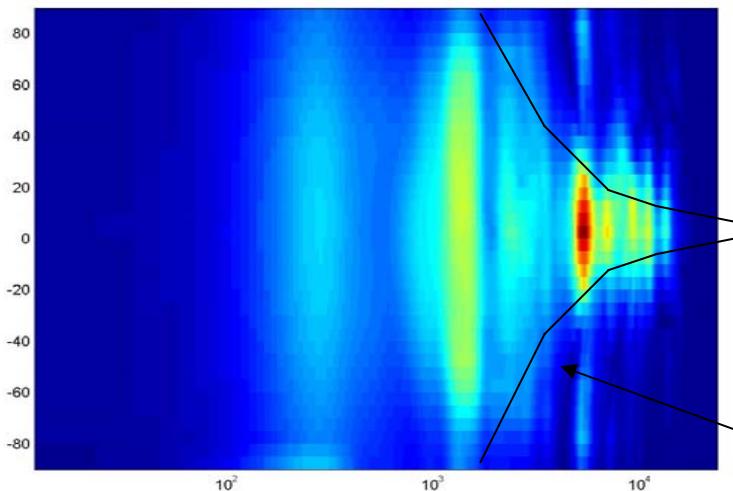
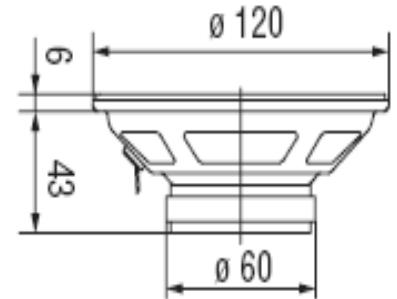
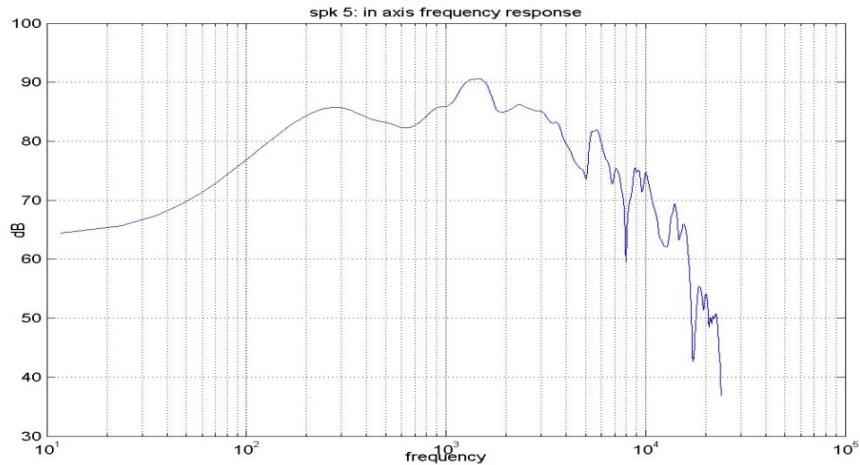


Omnidirectional



Rigid pistons

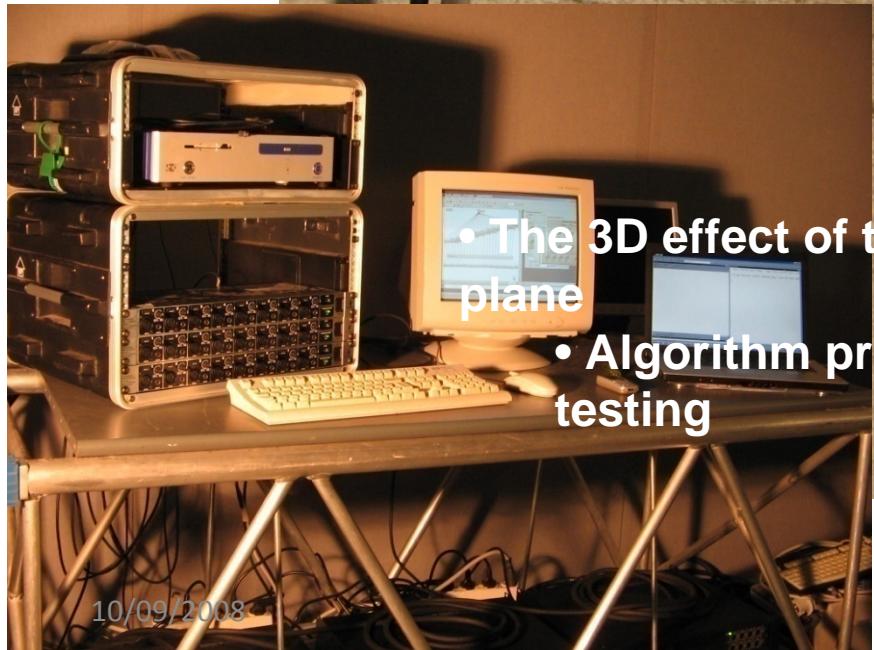
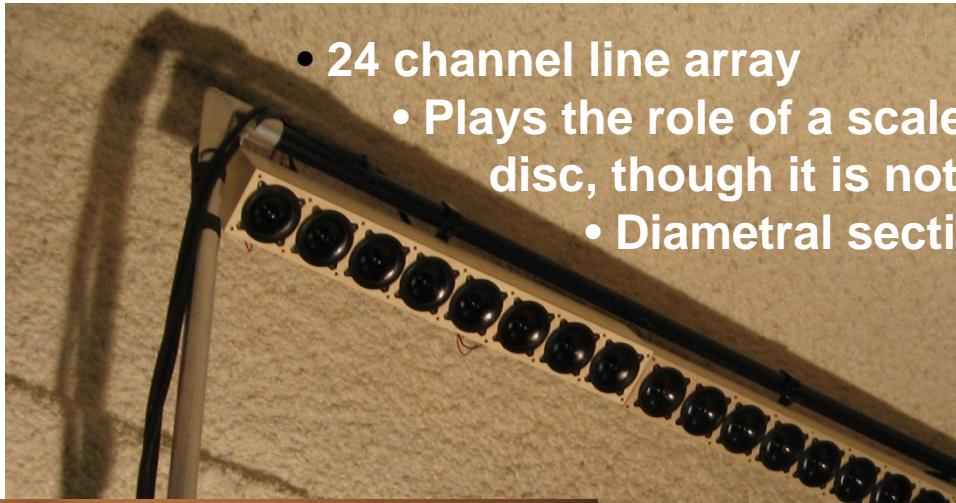
Our choice: single transducer, extended range



$$\operatorname{sen} \vartheta_{em}(f) \leq \frac{c}{2\Delta xf}$$

A simplified prototype

- 24 channel line array
- Plays the role of a scale model of the disc, though it is not
 - Diametral section of the disc



- The 3D effect of the disc is projected on a plane
- Algorithm production and testing, speaker testing

Chandellier: design, manufacturing and assembling

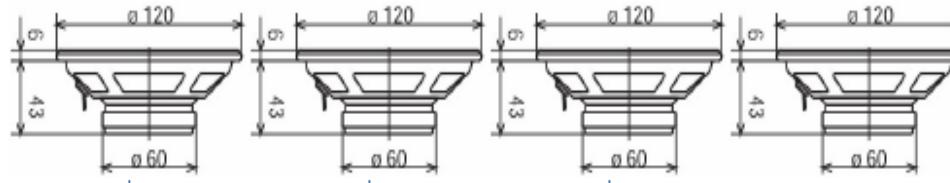
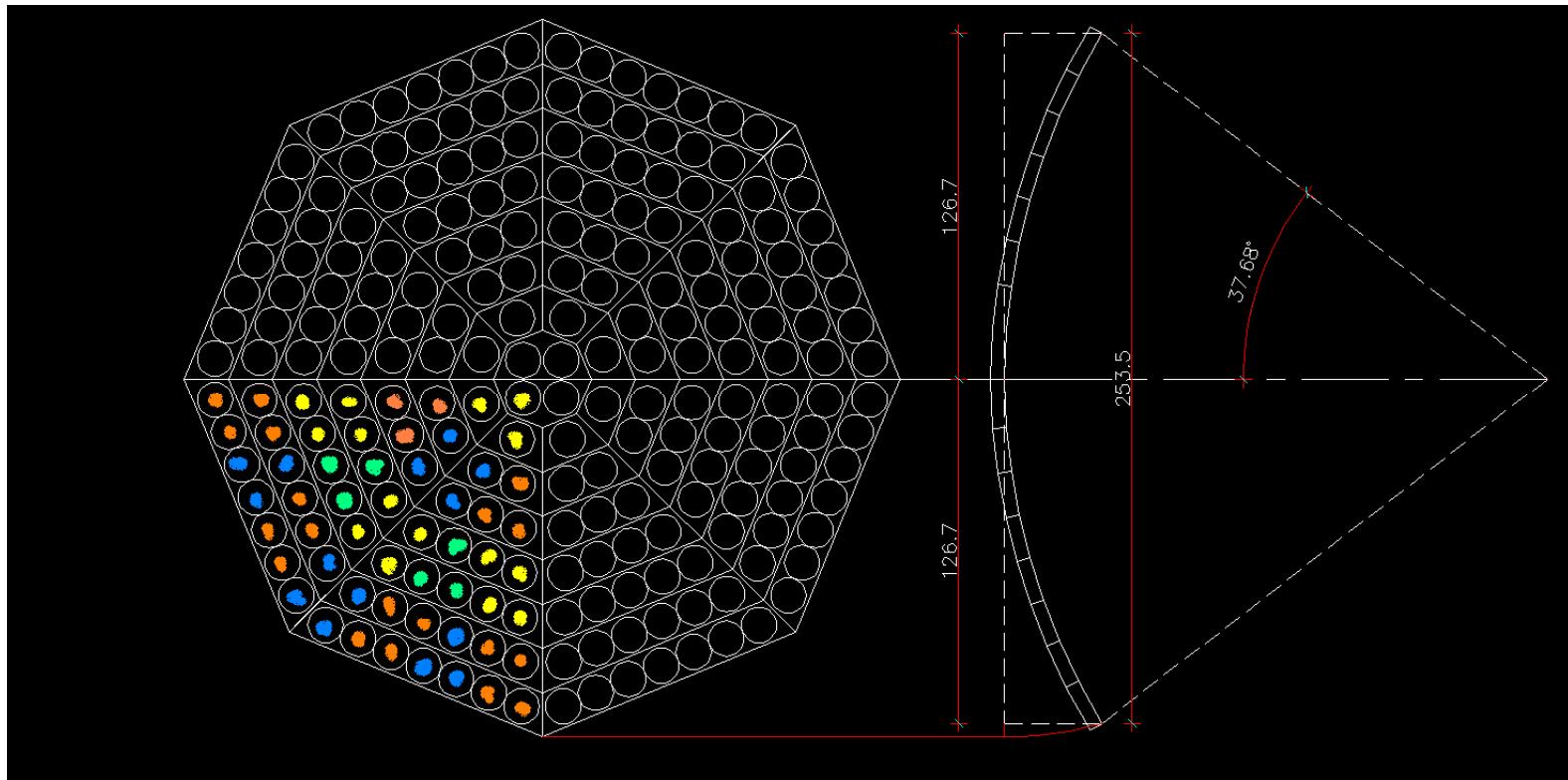


Lifting up..



10/09/2008

Speakers to channel connection



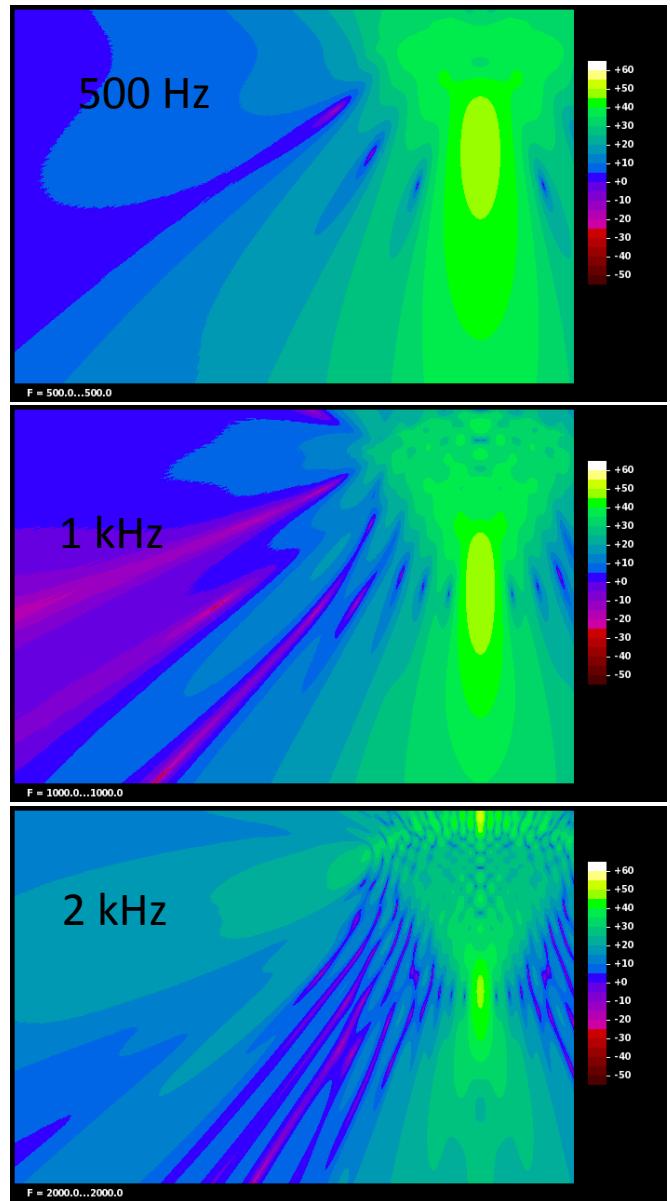
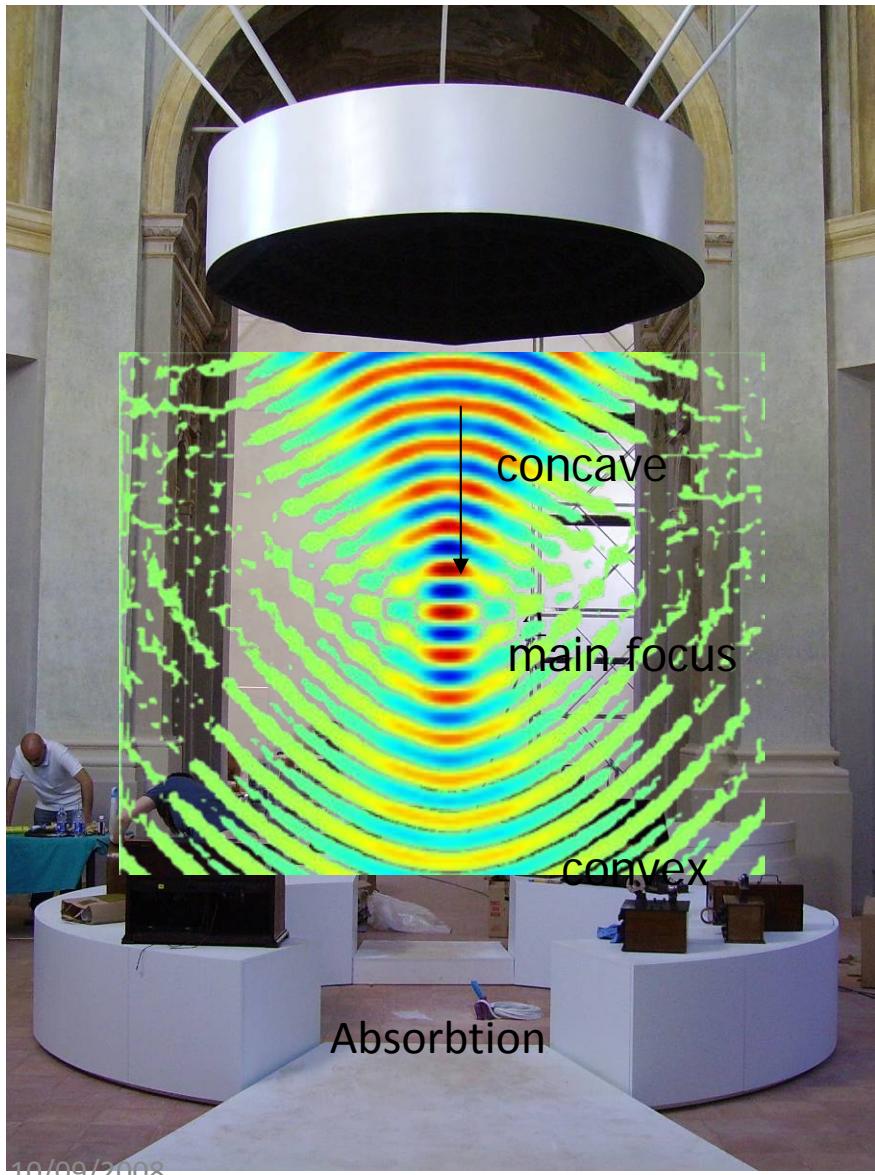
Special 32 Ohm
model by Ciare, Italy

228 loudspeakers
64 channels
10/09/2008

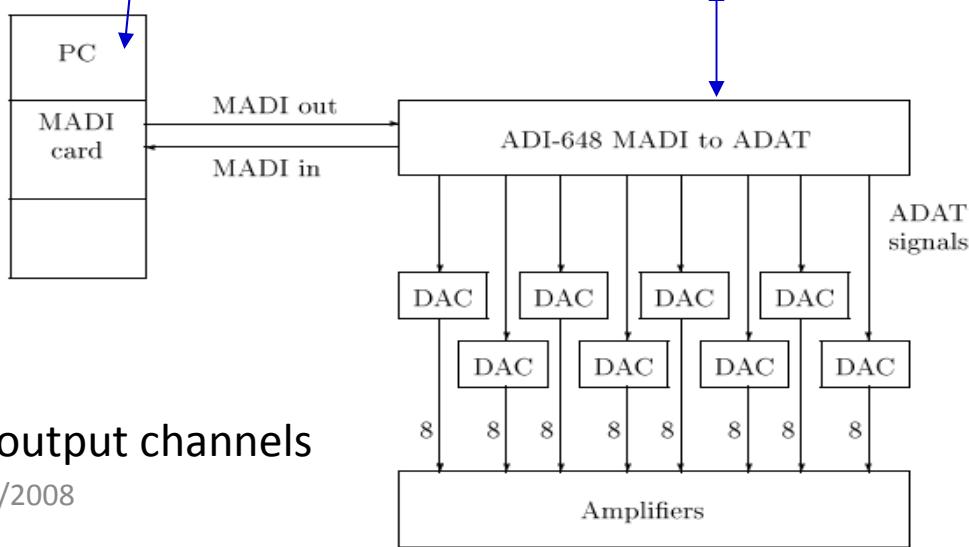
1 channel

Frounhofer distance for a group of speaker at 10 kHz is about **4** meters

More on the sound field..



The feeding system



From WFS to filter structure

Common WFS EQ
and speaker EQ
(mag and phase)

Antialiasing
sampling
filter

Gains and
delays

Antialiasing
reconstruction
filter

Single
speaker
relative EQ

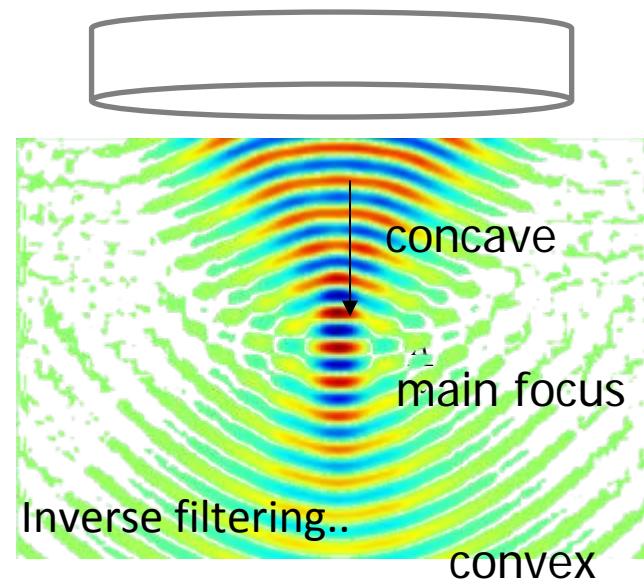
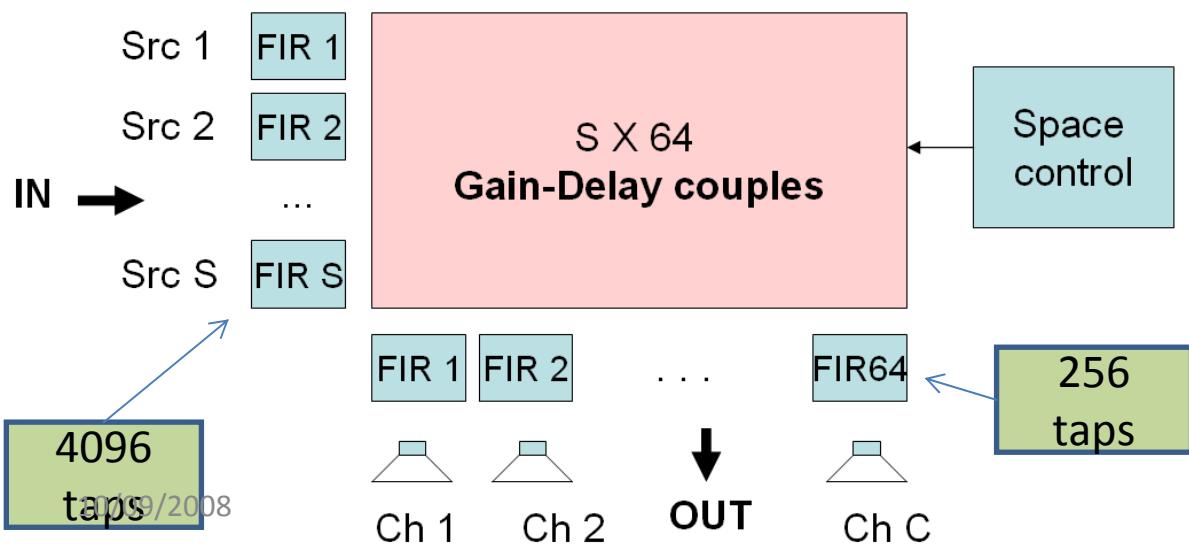
Input FIR

Array
shape

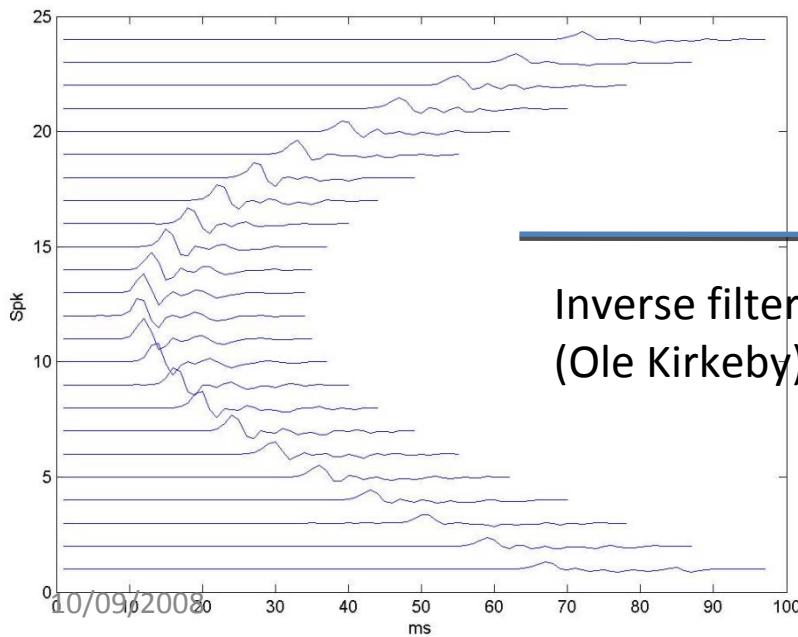
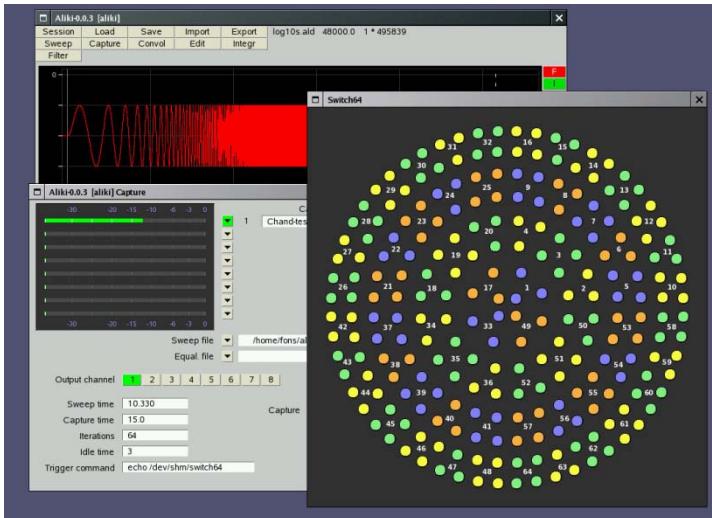
Dynamic
delays-gains
matrix ,
Output FIRs

Speaker
directivity

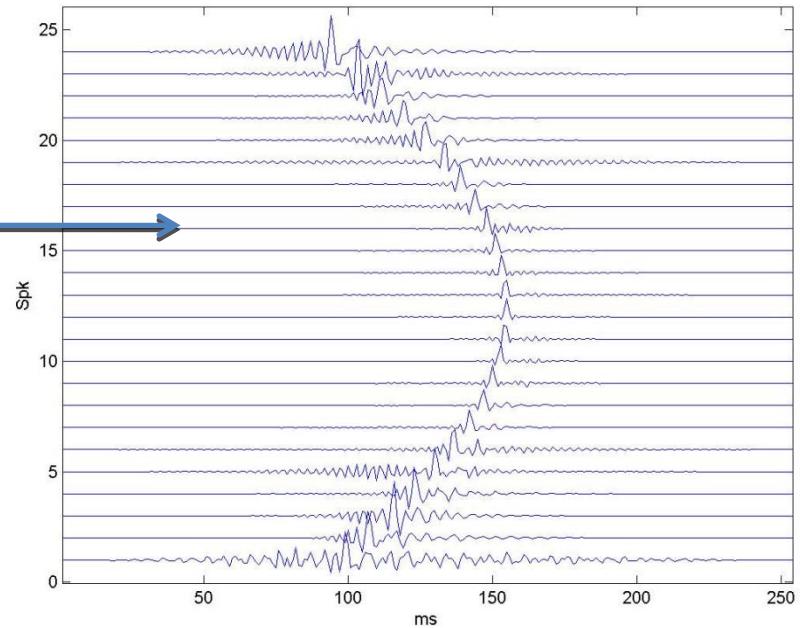
Output FIRs



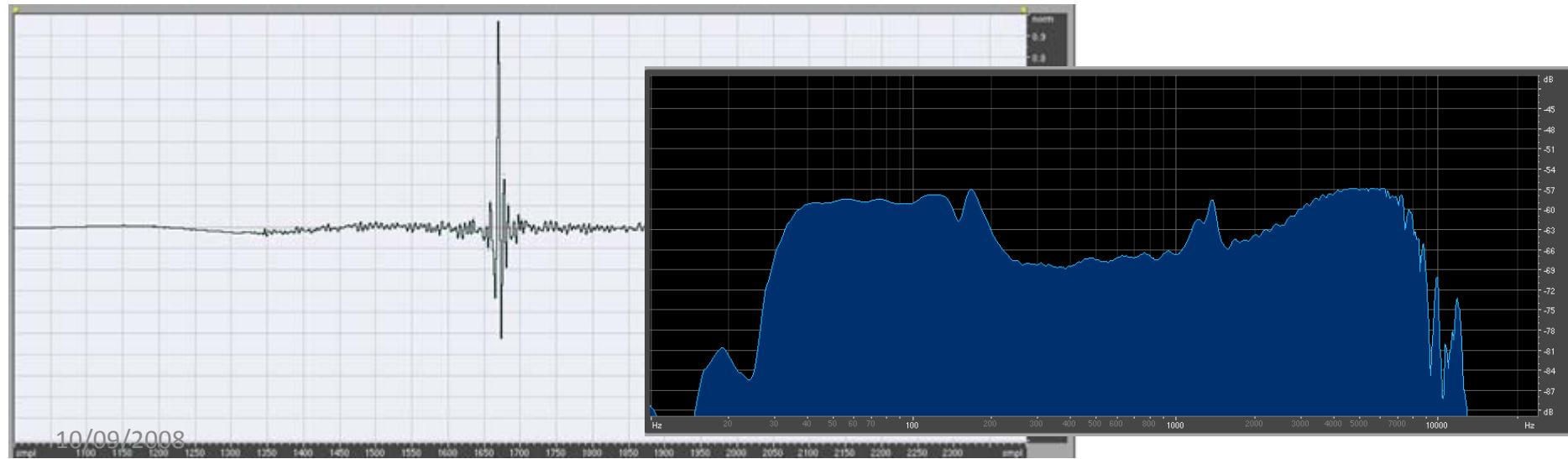
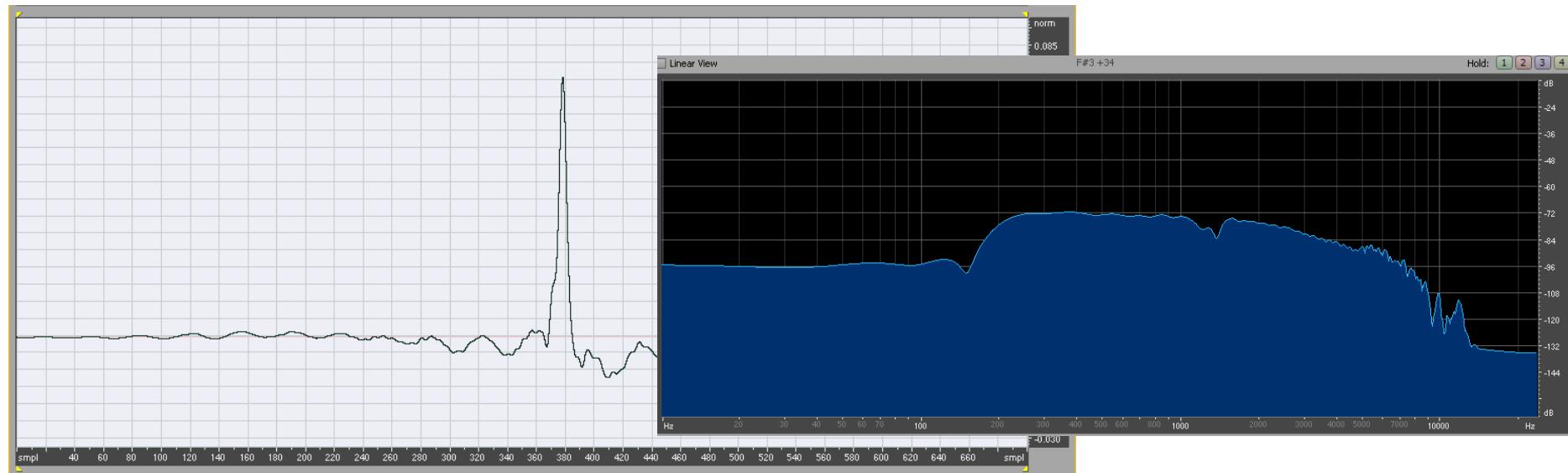
Output FIR design



Inverse filtering
(Ole Kirkeby)



Input FIR design

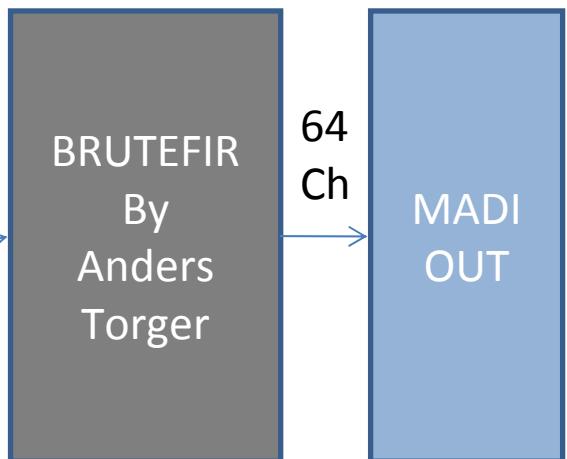
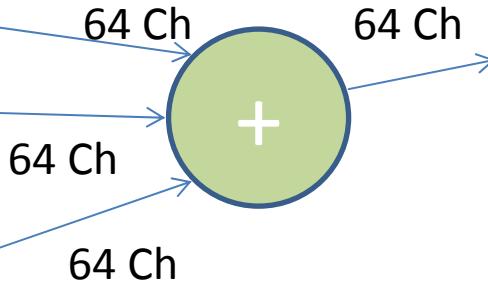


From filter to SW structure

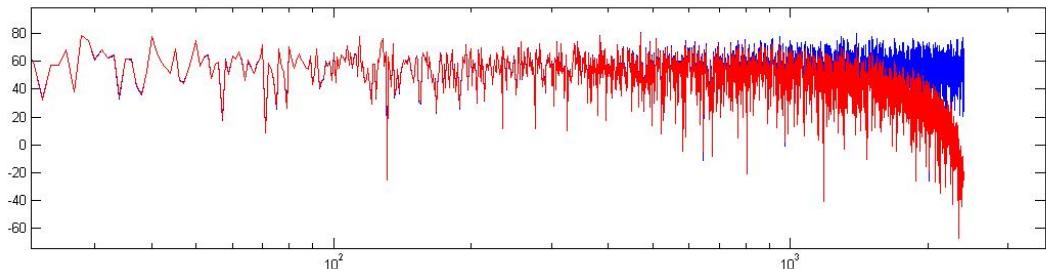
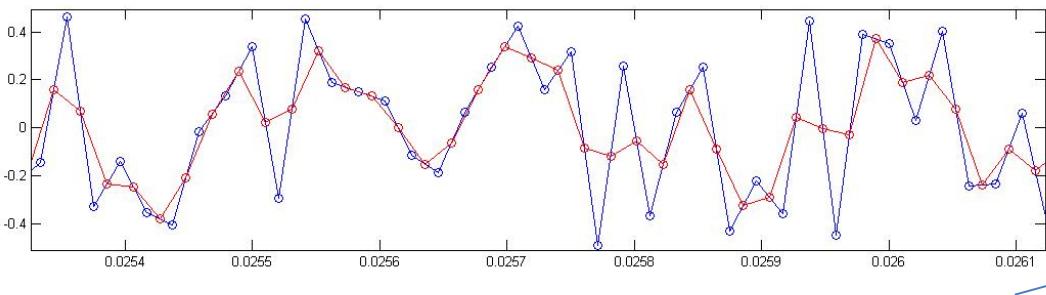
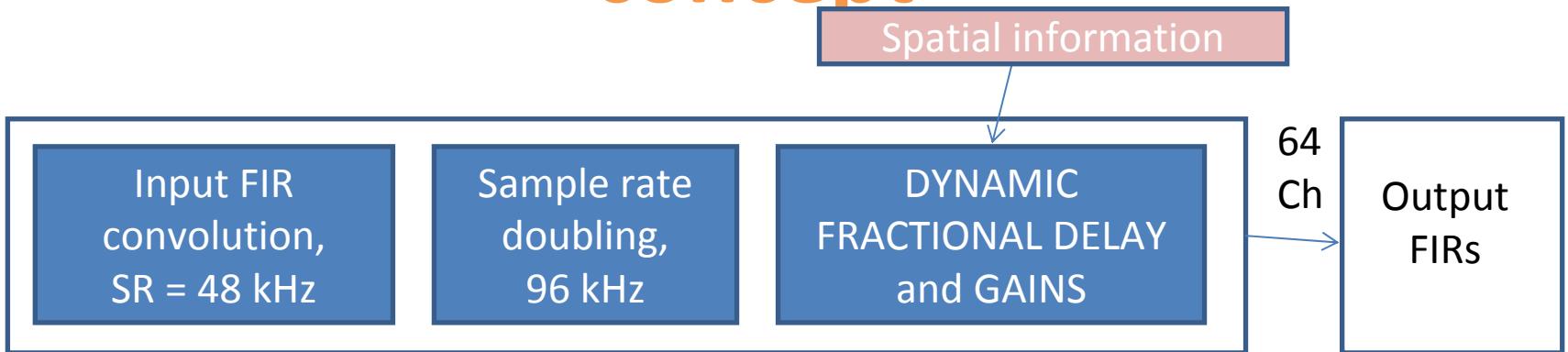


WFS focus:
By Daniele Torelli and
Fons Adriaensen

input FIR
dynamic gains and delays

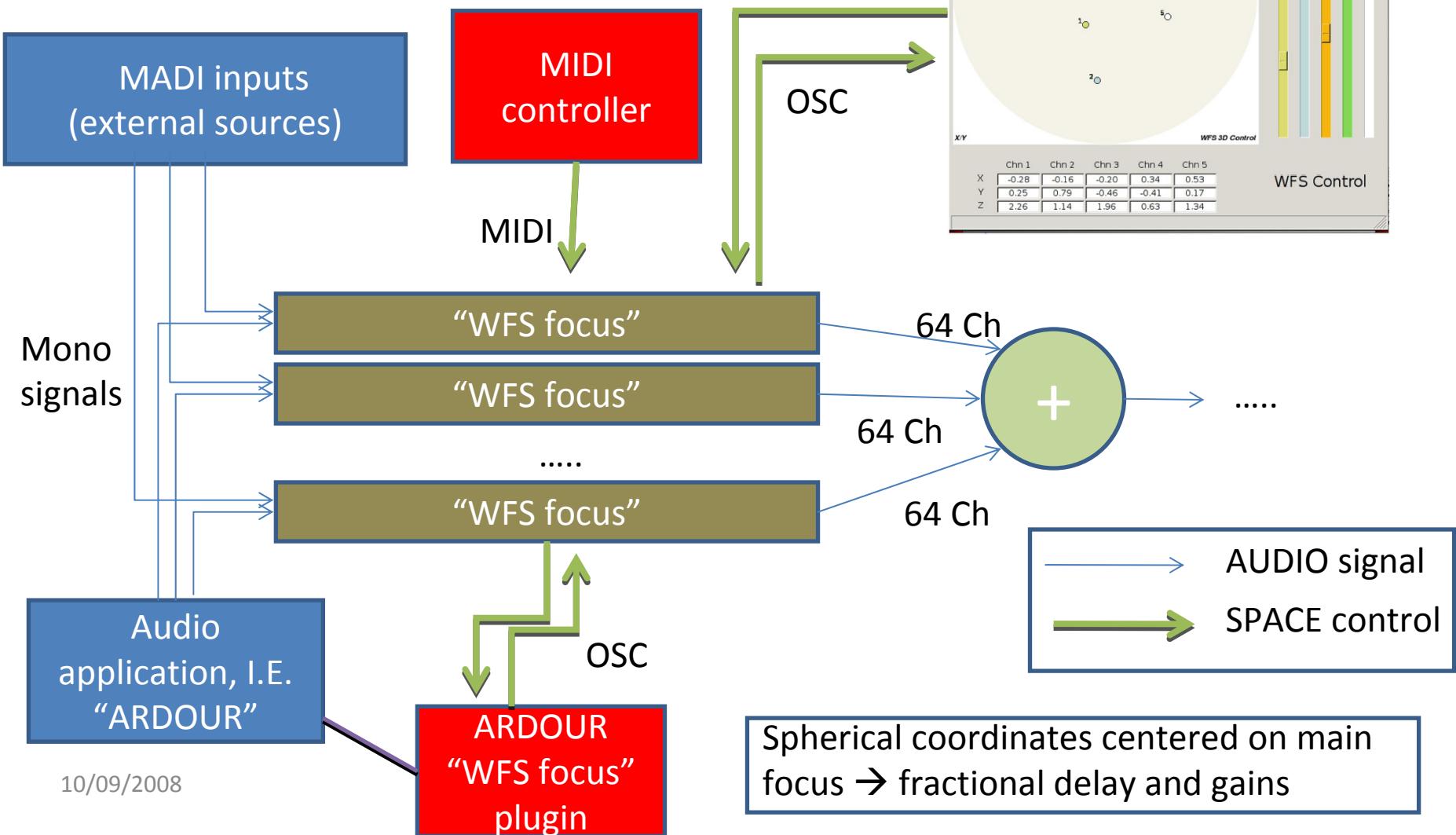


Moving the focus: “WFS focus” concept



Effect of a fractional delay with linear interpolation.

User interfaces



Results

- Precise and smooth, “click” free sound moving
- Appreciable Doppler effect
- “Objective” sound scene
- Good distance perception
- 22 dB decay at medium frequencies between the center and a peripheral point 5 m distant (center of one shell) . Tested with filtered pink noise.
- Very sensible increasing of the active insulation with people beneath the array

Thank you !