

Large-scale Higher Order Ambisonic Sound Reinforcement Systems

or:

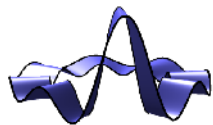
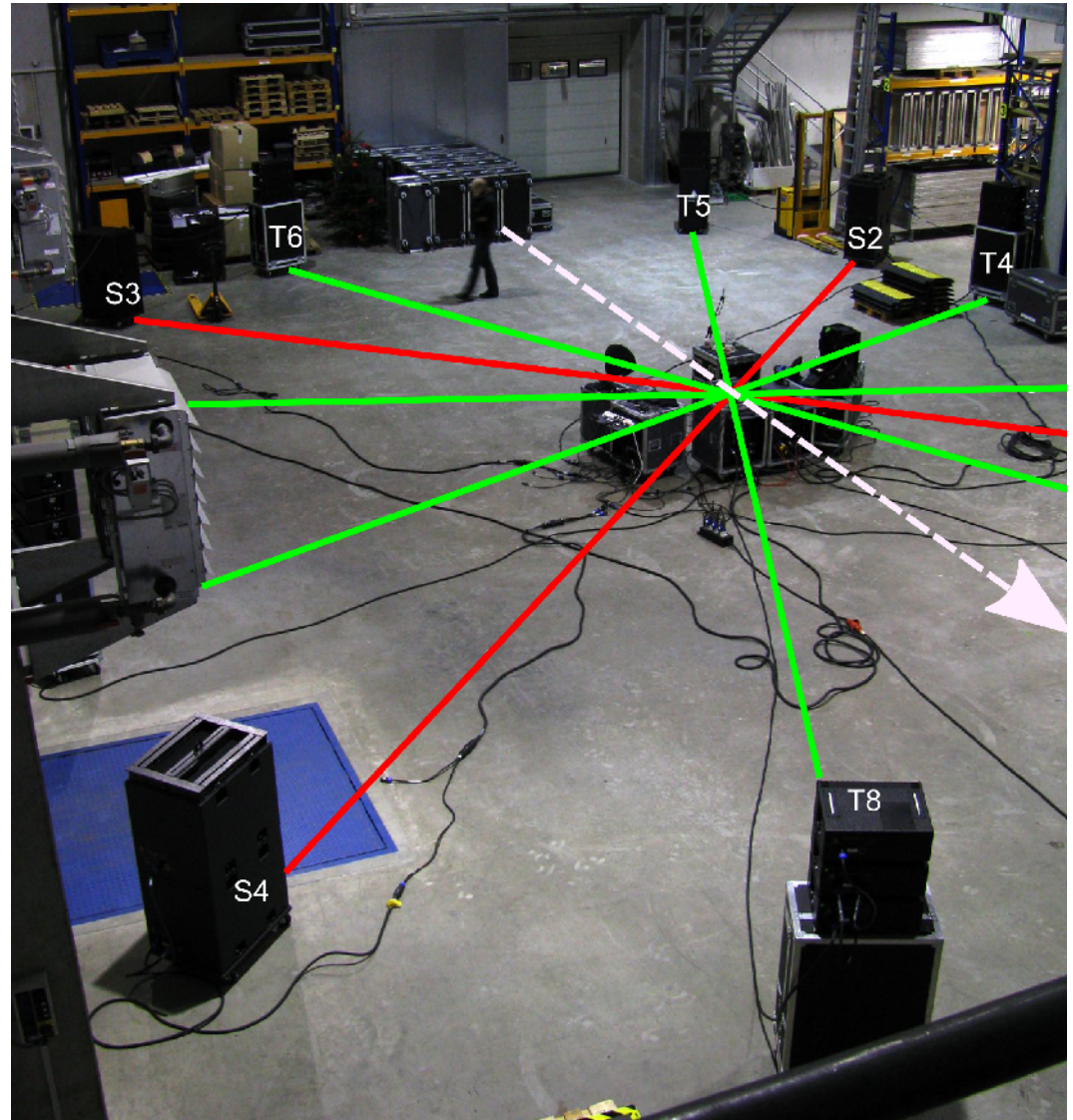
SIZE DOES MATTER

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Large-scale Higher Order Ambisonic Sound Reinforcement Systems

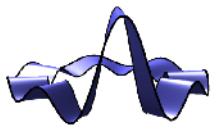
**SIZE
MATTERS
FOR**

- * large dance clubs
- * open-air dance events
- * rock opera shows

Large-scale Higher Order Ambisonic Sound Reinforcement Systems

Introduction:

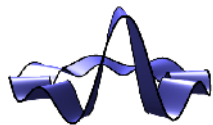
- Modern line arrays can be designed to deliver almost constant SPL over a wide area. Therefore, the only limiting constraint for large systems is run-time.
- Can we scale HOA systems to hundreds or thousands of listeners?
- What happens very far outside the sweet spot?



Large-scale Higher Order Ambisonic Sound Reinforcement Systems

Overview:

- Previous large-scale Ambi implementations
- Test system configuration
- Listening impressions
- Theoretical analysis
- Conclusion



Previous implementations:

Malham et al. 1990, “Mediamix 90”, York, UK:

8 horizontal speakers

4 elevated speakers

first-order decoder

max. radius: ca. 5m

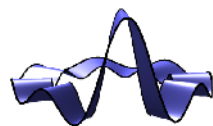
Malham et al. 1992, “York Mystery Plays”:

quasi-spherical: 1-4-6-4-1

first-order in-phase decode

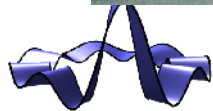
audience outside the sphere!

max. radius: ca. 7m



Previous implementations:

Wiggins/Funktion One 2006-2010, Glastonbury Festival, UK:
six horizontal
horn arrays
second-order
decode
max. radius:
~20m

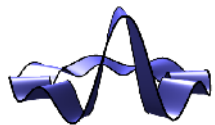
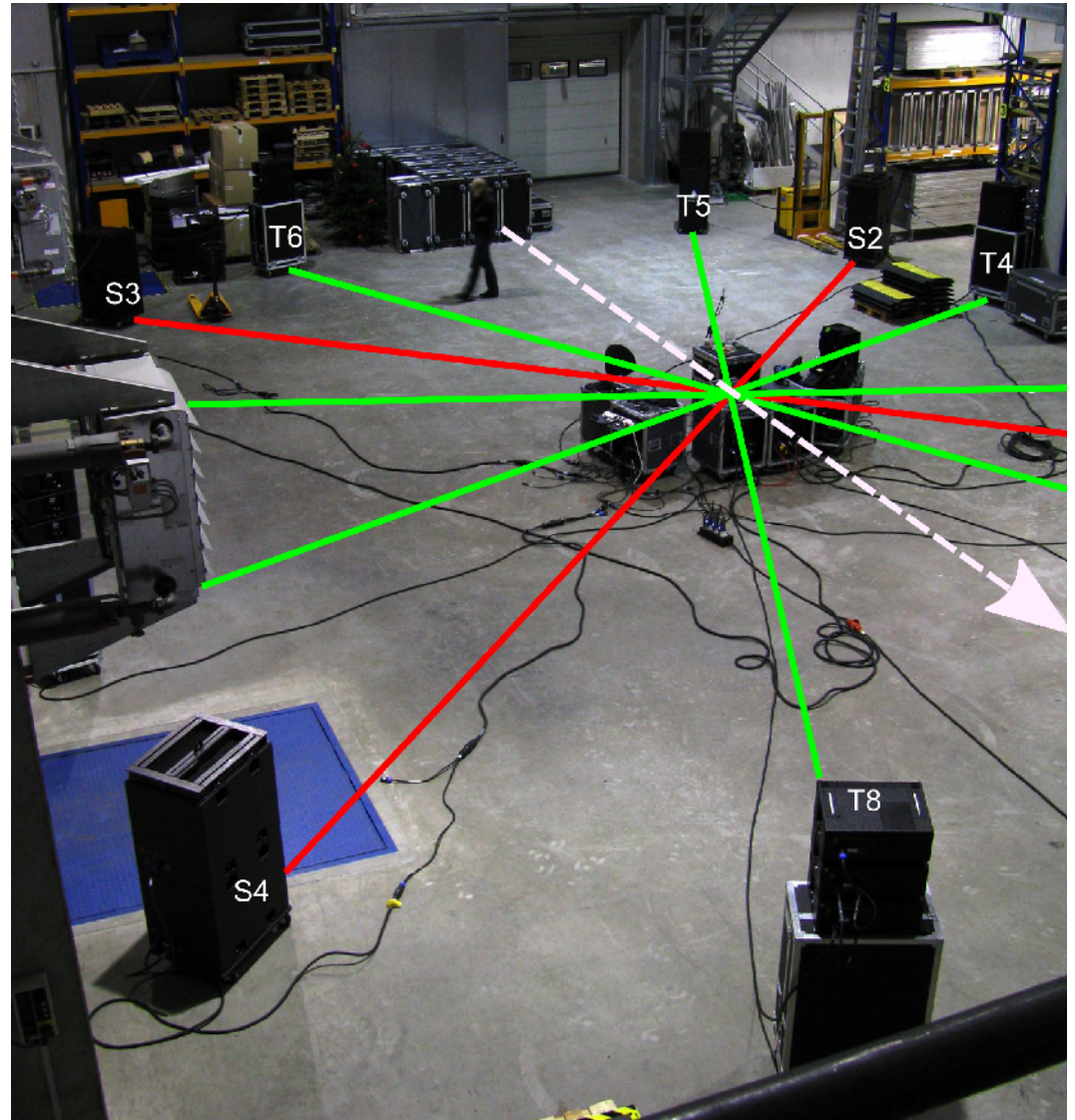


Configuration of the test system:

8 mid-hi stacks:
3x Adamson Metrix
8.5" LF, 3"->1.4" HF

4 sub stacks:
2x Metrix Sub
2x15"

X-over 100Hz,
24dB/Oct Linkwitz-
Riley

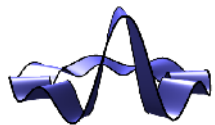
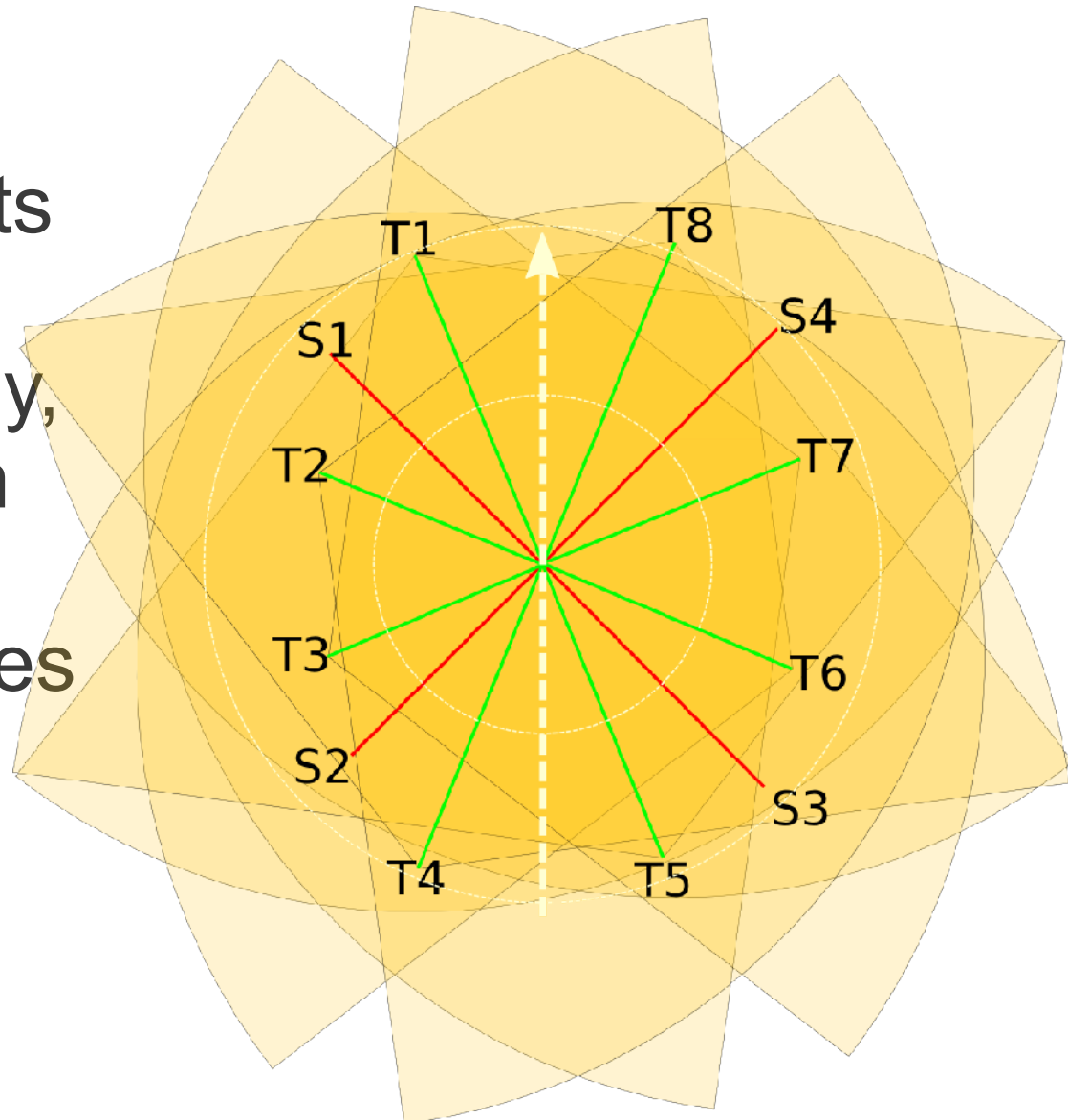


Configuration of the test system:

slightly oblong layout
due to space constraints

compensated with delay,
effective radius 10.25m

nominal coverage angles
mid-hi: 120°
subs: omni



Configuration of the test system:

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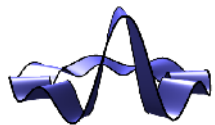
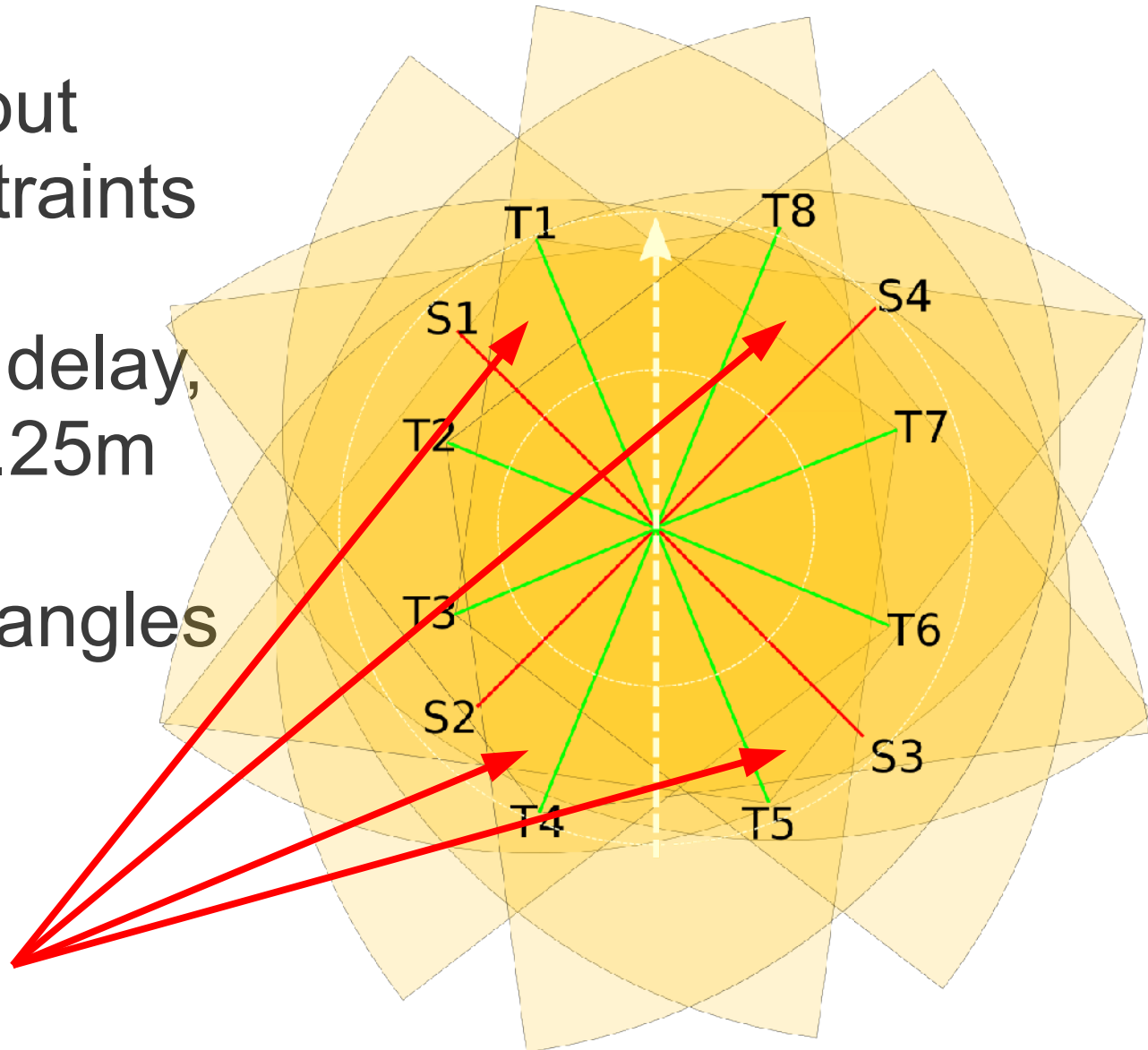
compensated with delay,
effective radius 10.25m

nominal coverage angles

mid-hi: 120°

subs: omni

coverage issues!



Configuration of the test system:

Mid-hi decoder: 3rd order,
max rV/max rE (x/o 500 Hz)

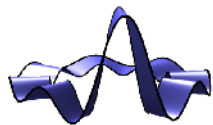
Sub decoder: 1st order,
max rV, later sub-cardioid

Decoder: AmbDec (Fons Adriaensen),
delay and near-field compensation switched on.

Gain calibration w/ pink noise using A-weighted
SPL meter at sweet spot, to +/- 0.1 dB

Listening impressions:

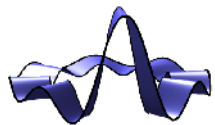
NB: Two subjects only (the authors)!



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Yeah, that's unscientific. But try to get people to attend a listening session two days before Christmas, with three days' notice, in a cozy warehouse heated to 12°, from 20-24:00h.



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Listening impressions:

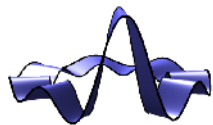
First-order material:

ok'ish for background music, not focused enough to draw attention to stage or performers

Third-order material:

stable and focused over a wide area; tested with single sources and hybrid HOA recordings

very large perceived source size - irritating with nature recordings, but kind of nice for rock shows
← not specific to Ambisonics



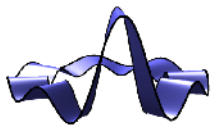
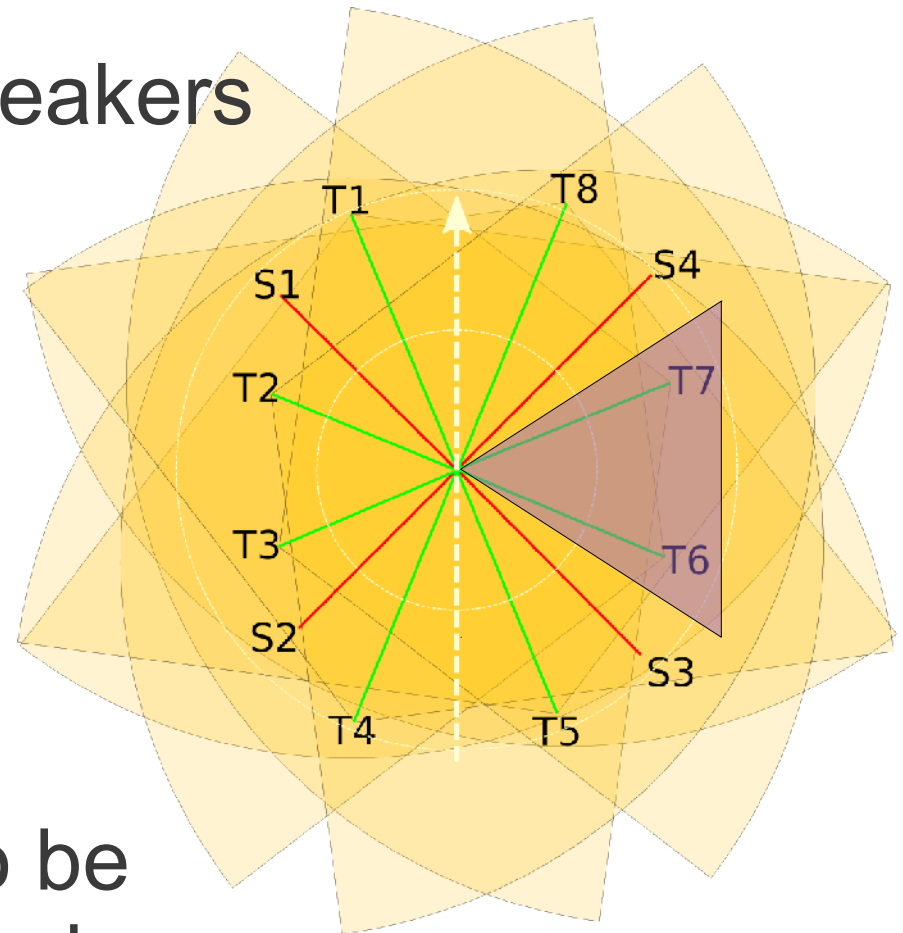
Listening impressions:

A/B comparison: native Stereo vs. virtual speakers

Base angle 45°

Stacks T7 and T6
with S4 and S3 used
for symmetry reasons.

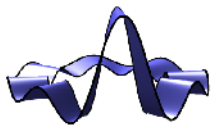
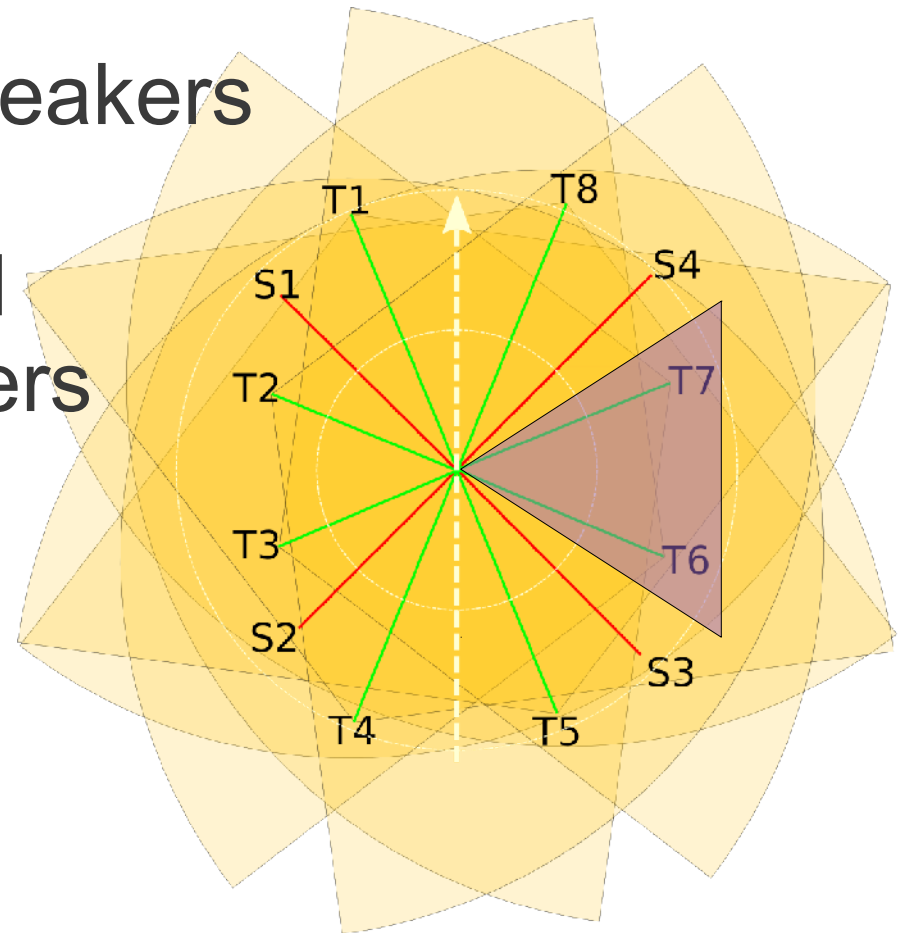
Virtual speakers panned to be congruent with physical stacks.



Listening impressions:

A/B comparison: native Stereo vs. virtual speakers

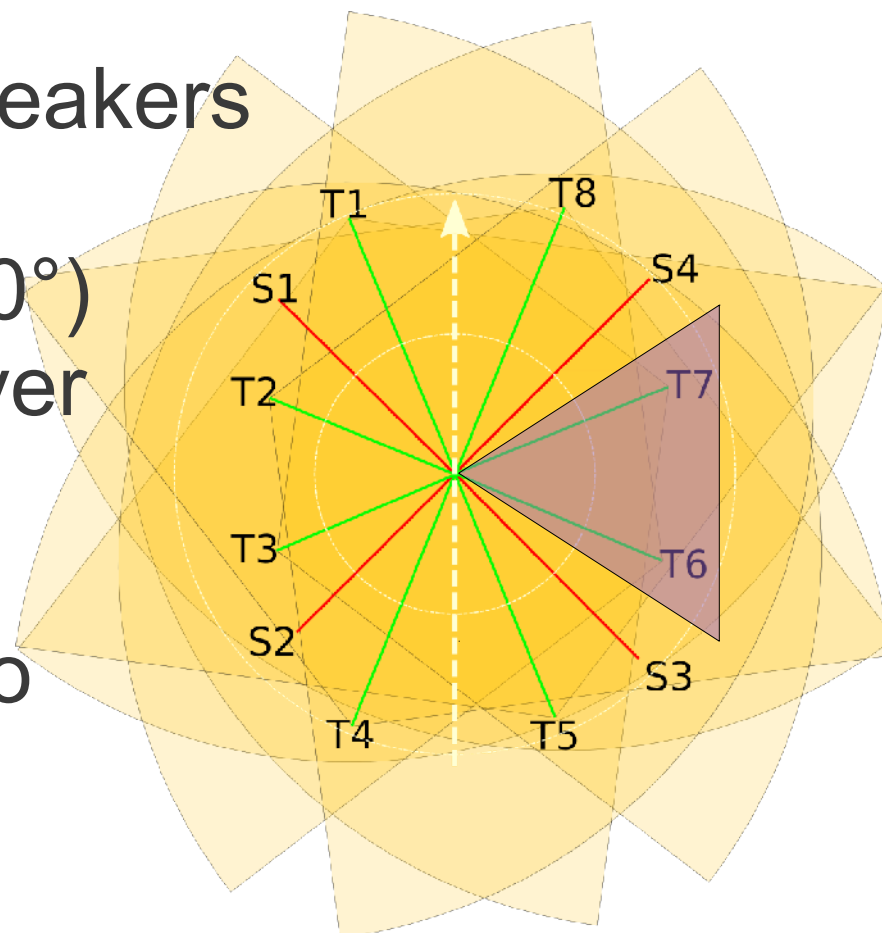
slight loss of presence and dryness with virtual speakers in direct comparison;
still very enjoyable



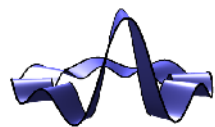
Listening impressions:

A/B comparison:
native Stereo vs. virtual speakers

wider virtual stereo (90-120°)
preferred by both testers over
native stereo;
very immersive, no “hole-
in-the-middle” effect, stereo
localization remains stable



→ an improvement even for standard stereo material



Listening impressions:

Sub decode:

Max rV means out-of-phase components:

→ loss of “punch”, less efficient

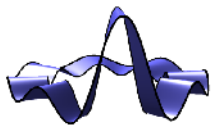
→ no perceptible gain in localization



Sub-cardioid or even mono decode is better suited to rock'n'roll settings, unless

→ the crossover frequency is very high, or

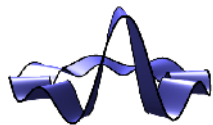
→ the system is installed in a small room (less excitation of room modes with hypercard decode).



Listening impressions:

- * very good localization over a wide area
 - * no evident coloration
 - * only very subtle phasing artifacts
 - * no collapse until within 1-1.5m of the opposite speaker even with hypercardioid decode
- } possibly masked by room reverberation!
- * confirmed usable area: 180m²
 - 360 tickets for dance floor, easily more if you provide lounge area around the listening zone.

We are confident that 600 people will have a satisfactory listening experience with adequate localisation.



Listening impressions:

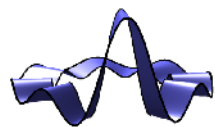
- * very good localization over a wide area
- * no evident coloration
- * only very little phasing artifacts } possibly masked by room reverberation!
- * no collapse with 11-speaker system
- * even with hypercardioid decode

Slight problem:

- * confirmed usable area: 180m²

→ 360 tickets for dance floor, easily more if you provide your own headphones
This is too good to be true.

We are confident that 500 people will have a satisfactory listening experience with adequate localisation.

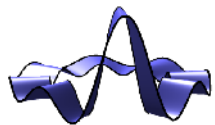


Theoretical considerations:

Almost everybody is outside the sweet spot.
Therefore, the **ITD** cues will be incorrect.

As the listener moves away from the center, the
ITDs will shift out of phase.

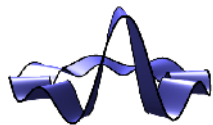
It seems those inconsistent cues are being
discarded by the brain and do not cause
noticeable confusion, as long as robust **ILD** cues
are available.



Theoretical considerations:

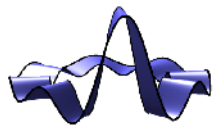
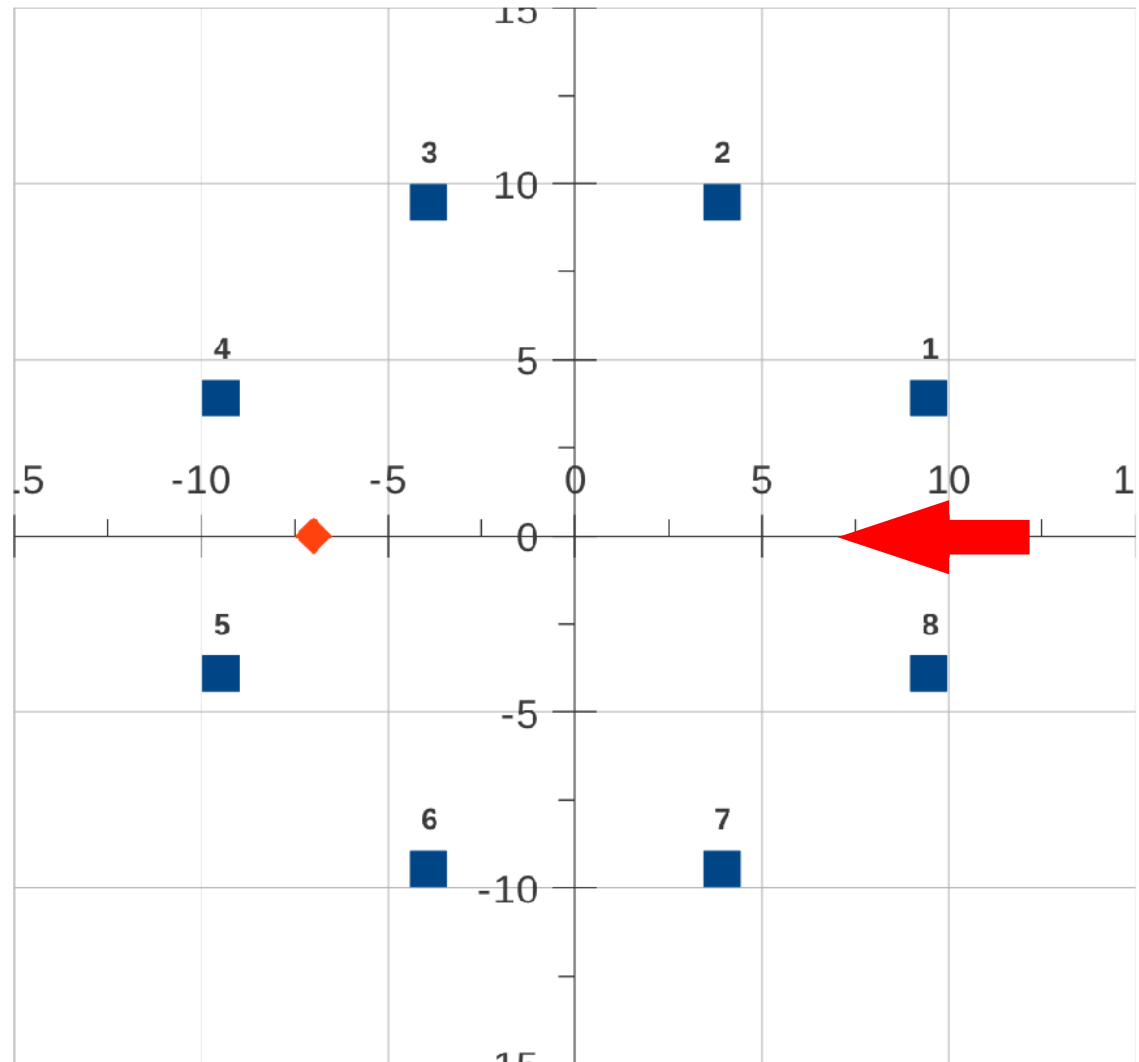
Our best metric is therefore rE .

Consequently, it might be best to use a max rE decode throughout the spectrum, perhaps excluding the subs (not tested).



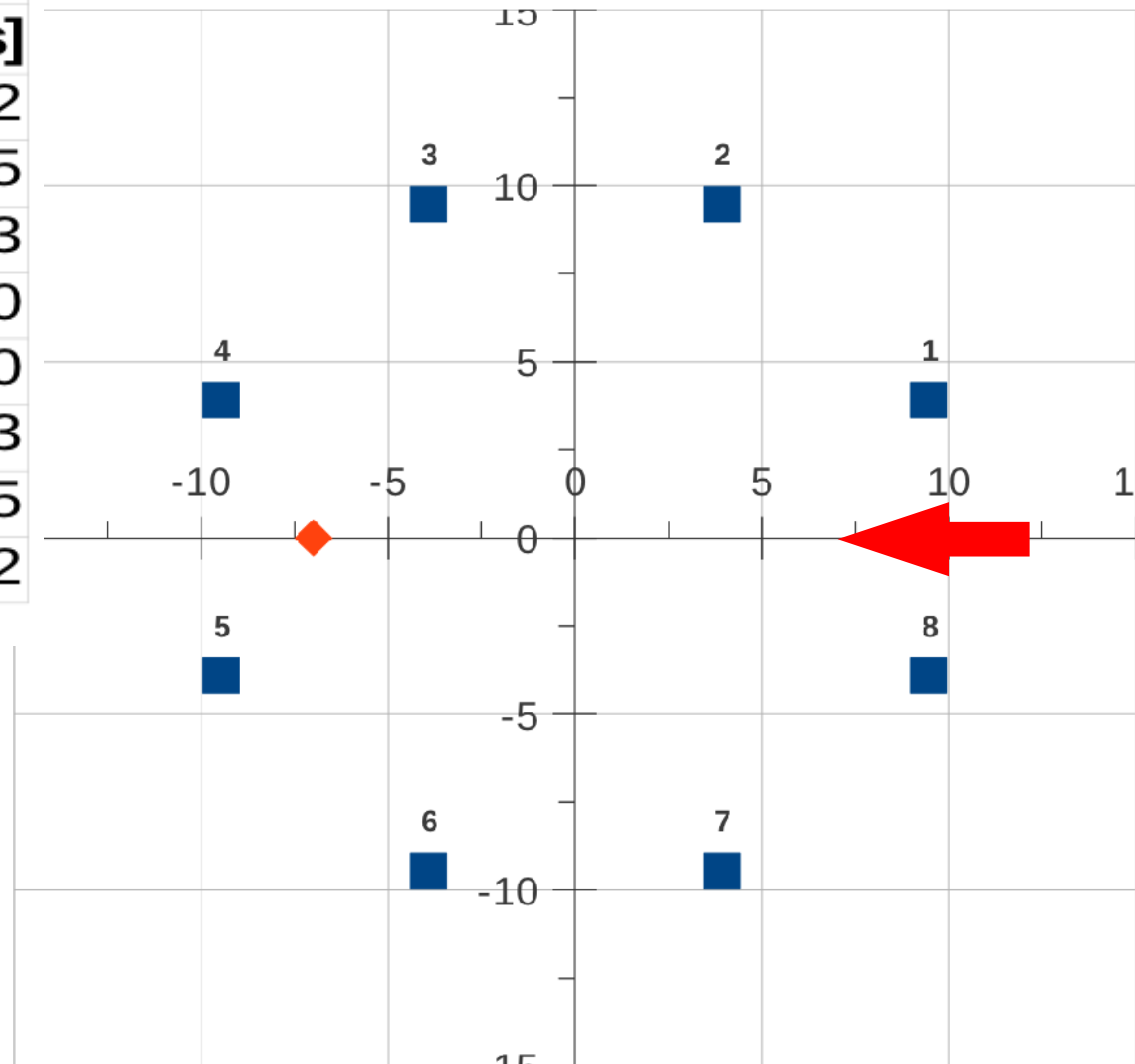
Theoretical considerations:

Worst-case scenario:
a source from 0° ,
listener at 180° ,
7m away from center

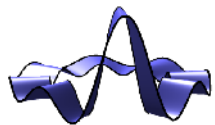


Theoretical considerations:

	relative level	relative delay
i	$\Delta p(i)$ [dB]	$\Delta t(i)$ [ms]
1	0.000	36.162
2	-10.000	28.885
3	-15.000	15.653
4	-18.000	0.000
5	-18.000	0.000
6	-15.000	15.653
7	-10.000	28.885
8	0.000	36.162

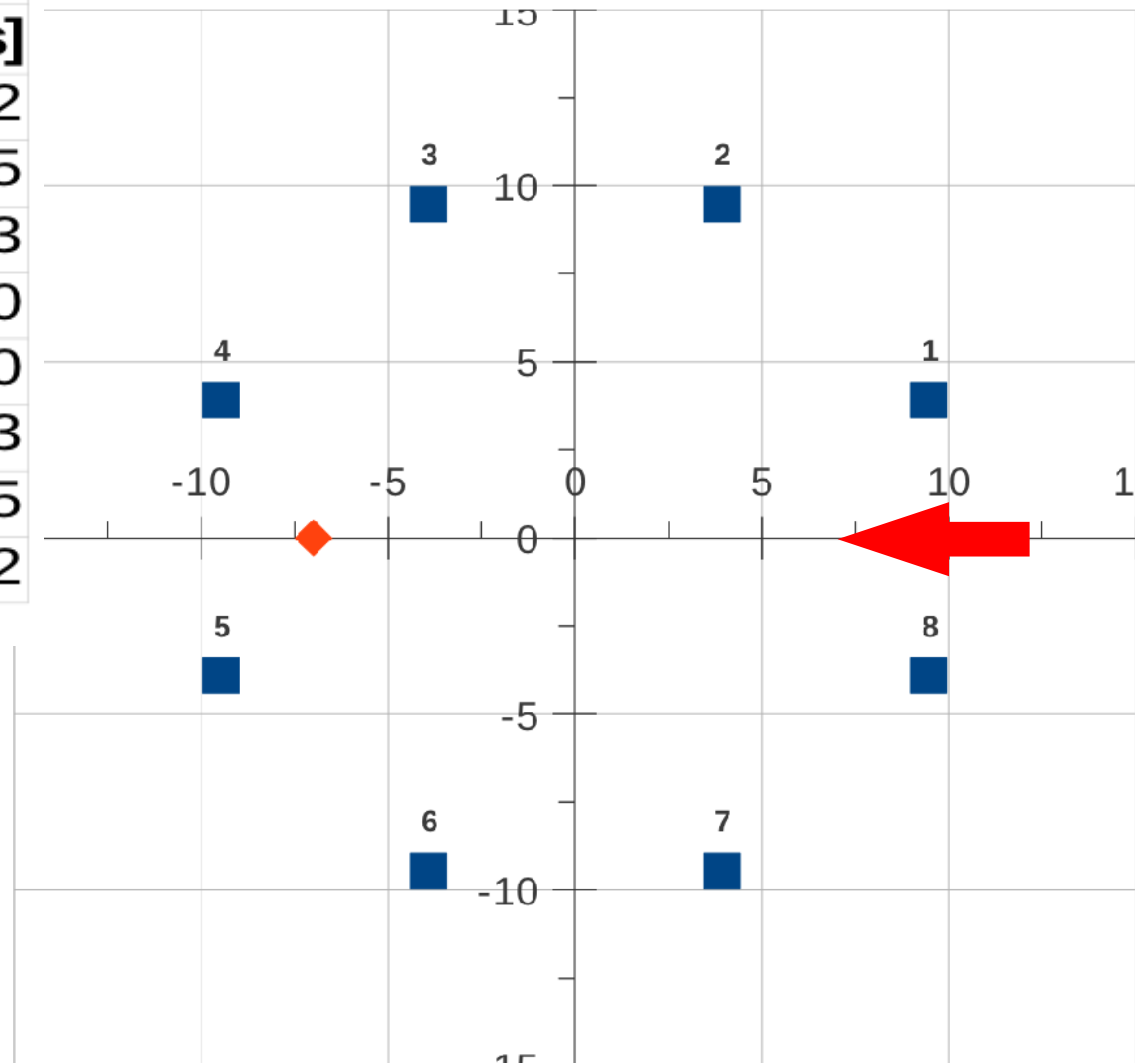


relative levels and timings
at listening position
(assuming constant SPL,
normalized to first
incoming wave front)

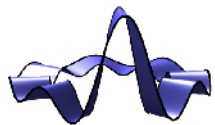


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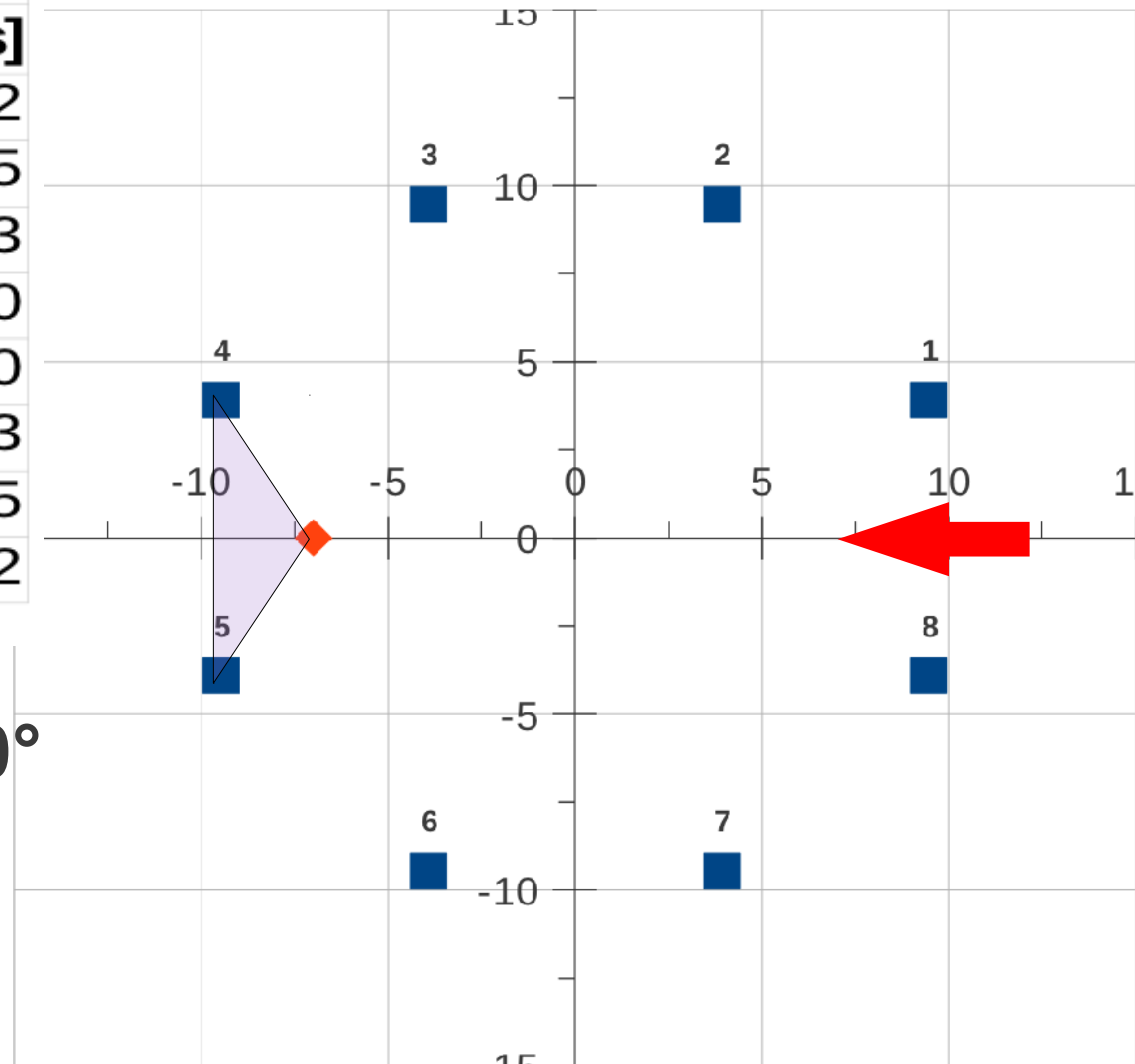


Applying stereophonic theory (summing localization, precedence effect, echo threshold):

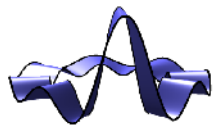


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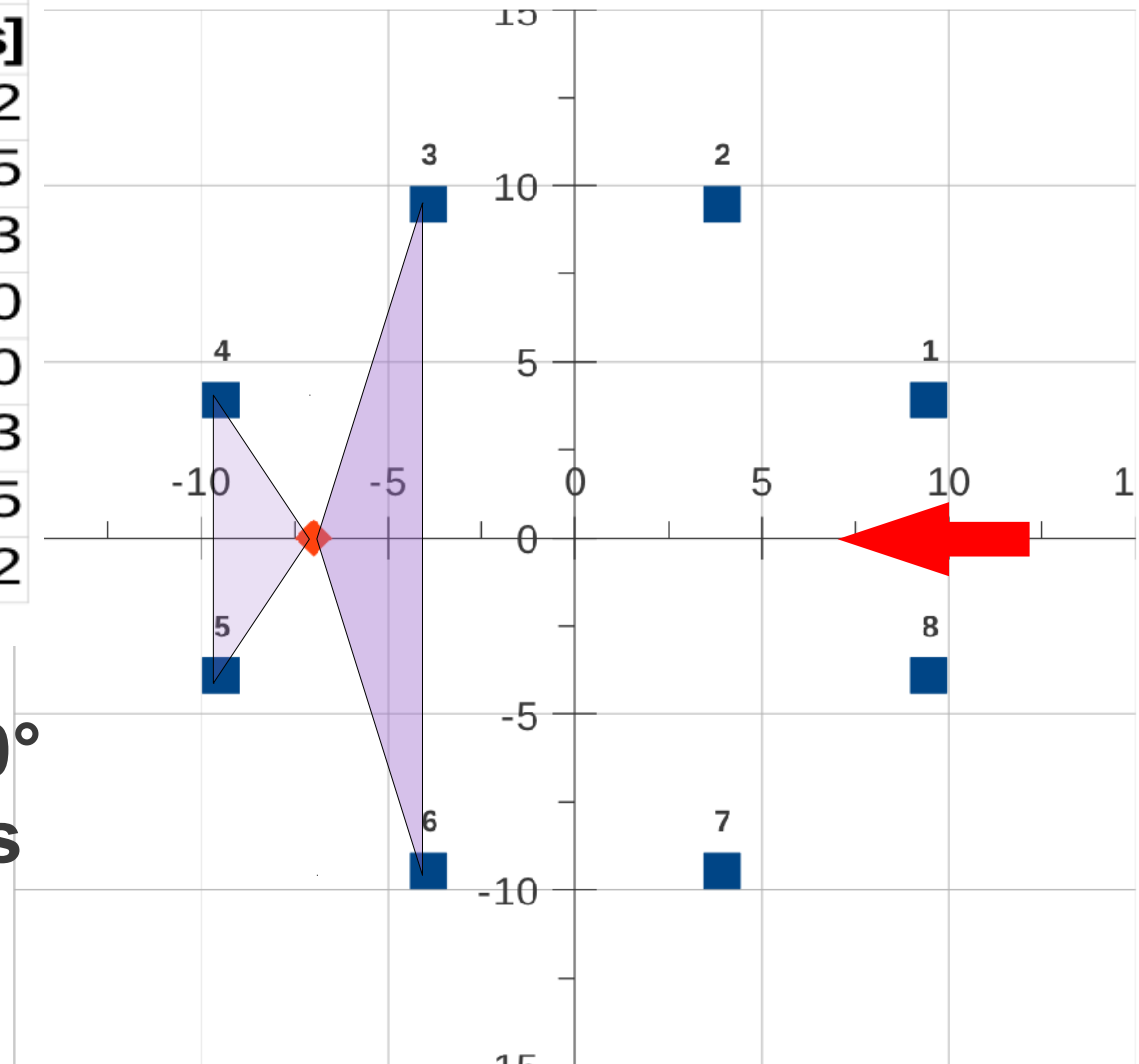


1st phantom source at 180°

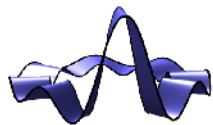


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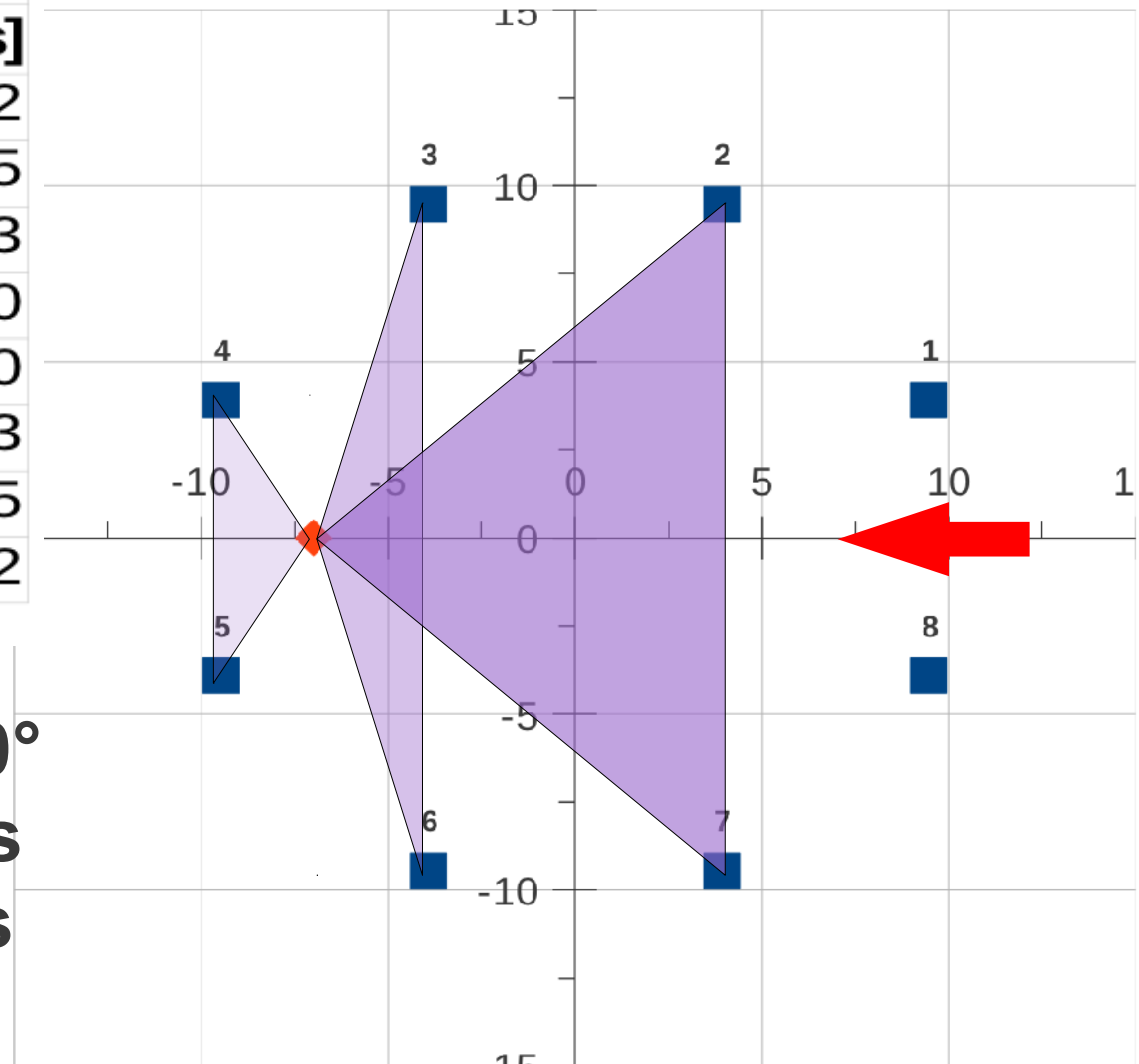


1st phantom source at 180°
 2nd p.s. at 0°, +3dB, +16ms

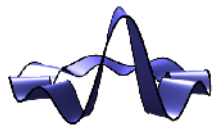


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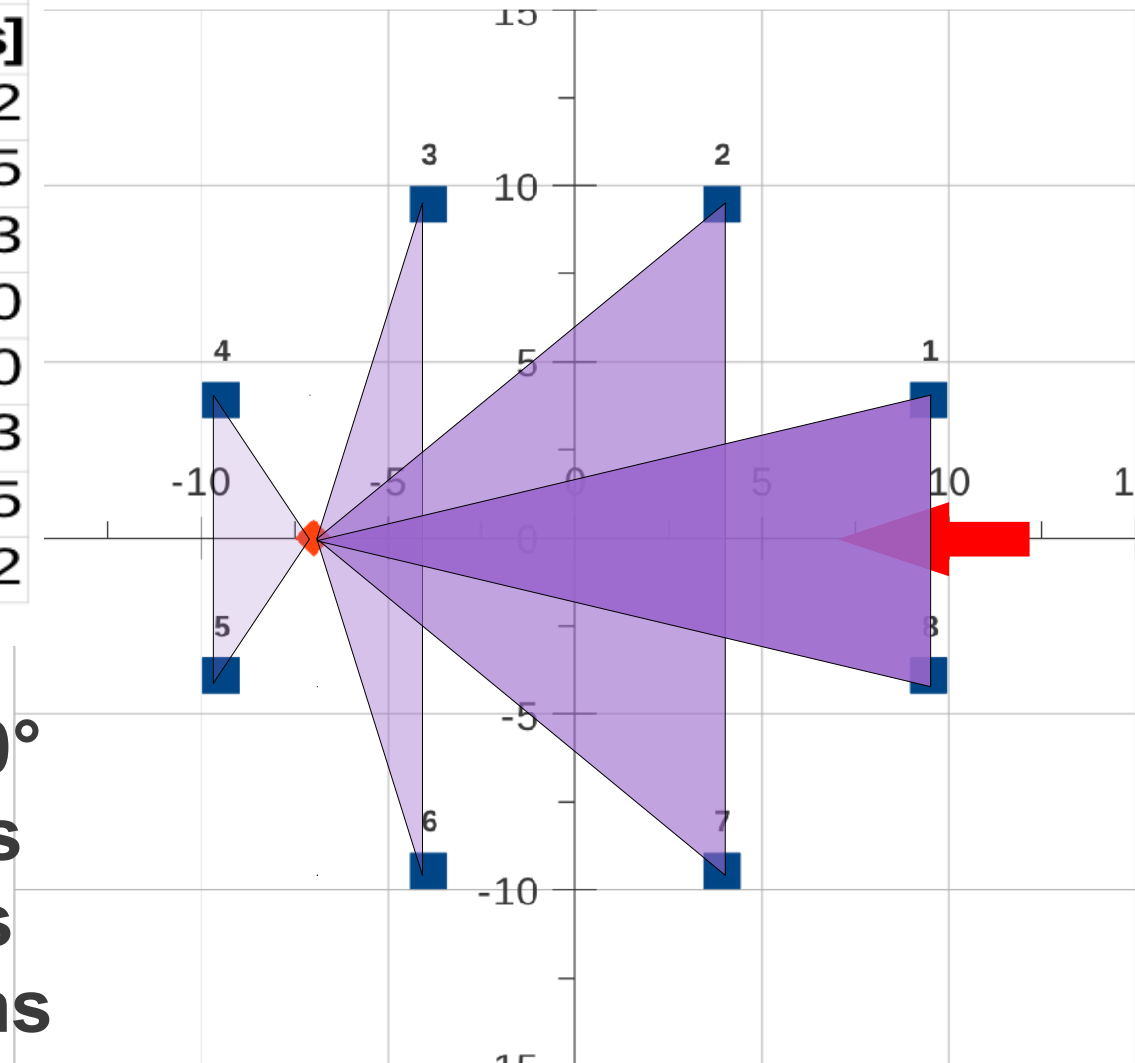


1st phantom source at 180°
2nd p.s. at 0°, +3dB, +16ms
3rd p.s. at 0°, +8dB, +29ms

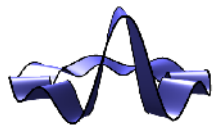


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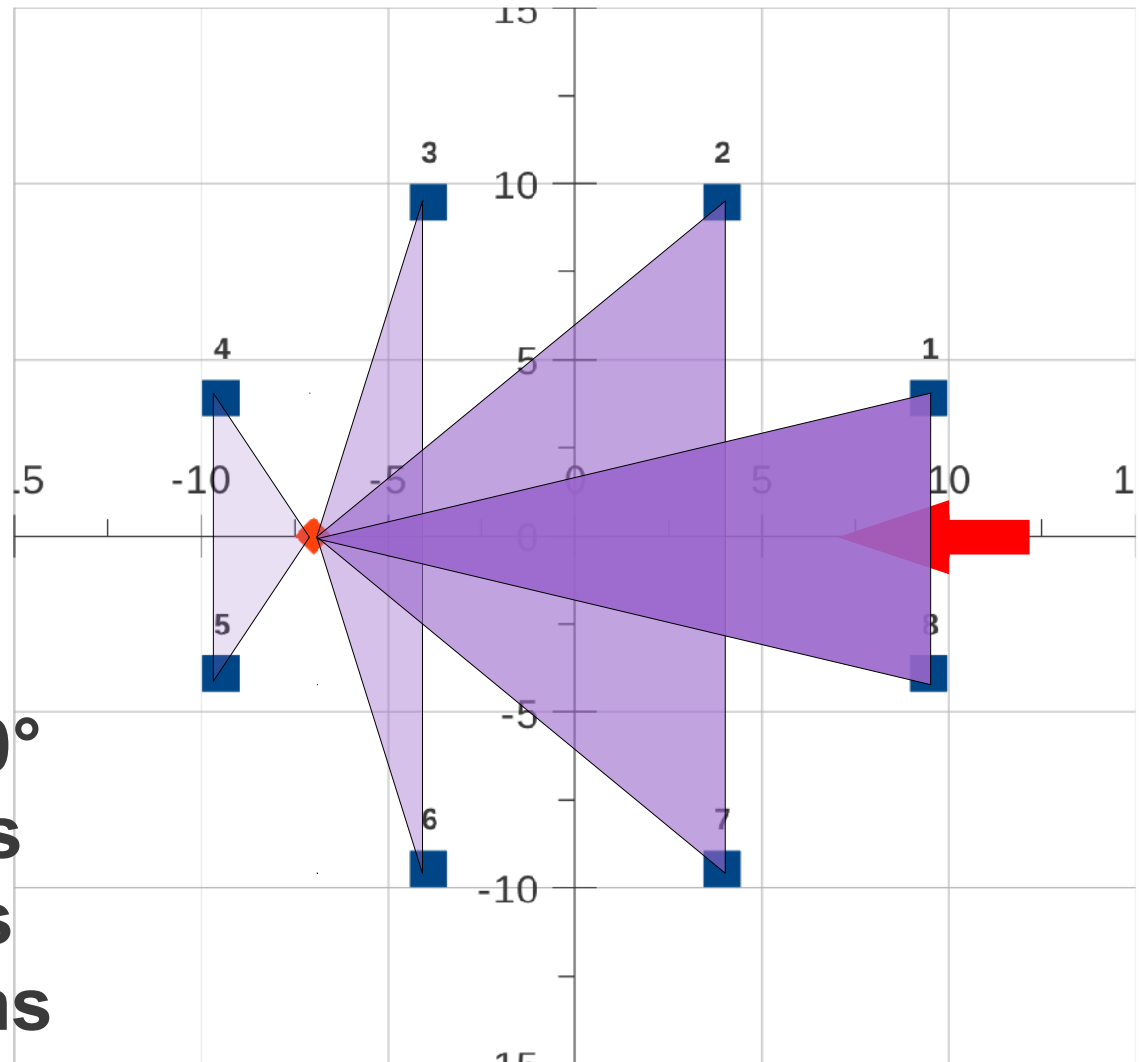
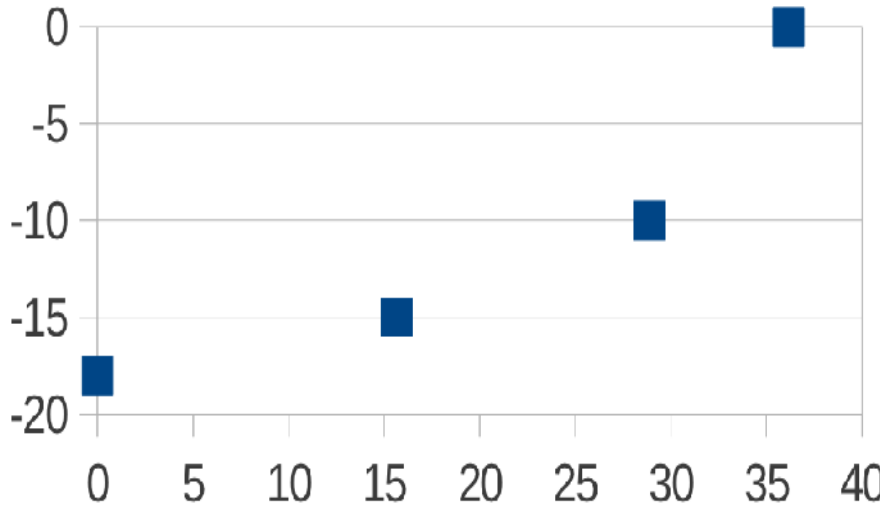
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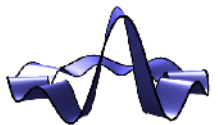
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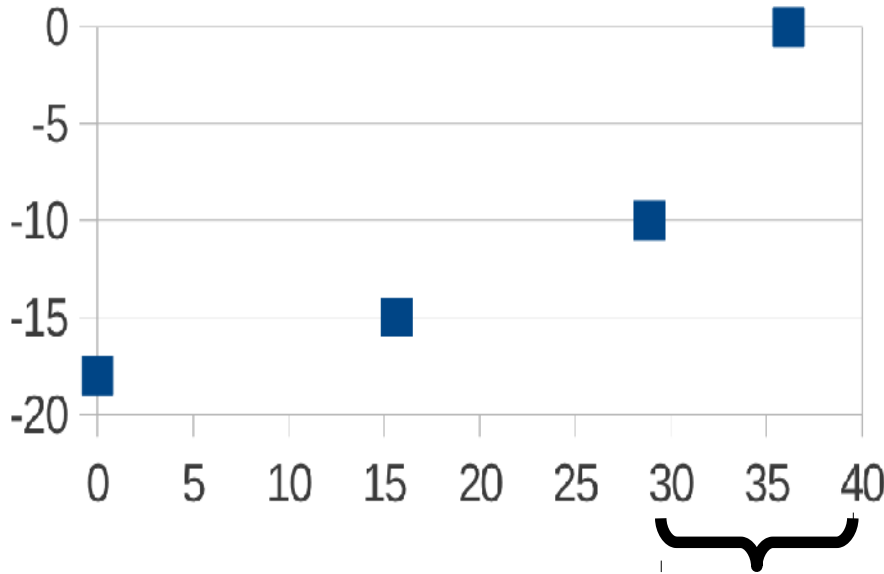
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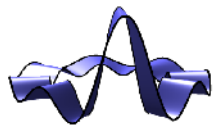
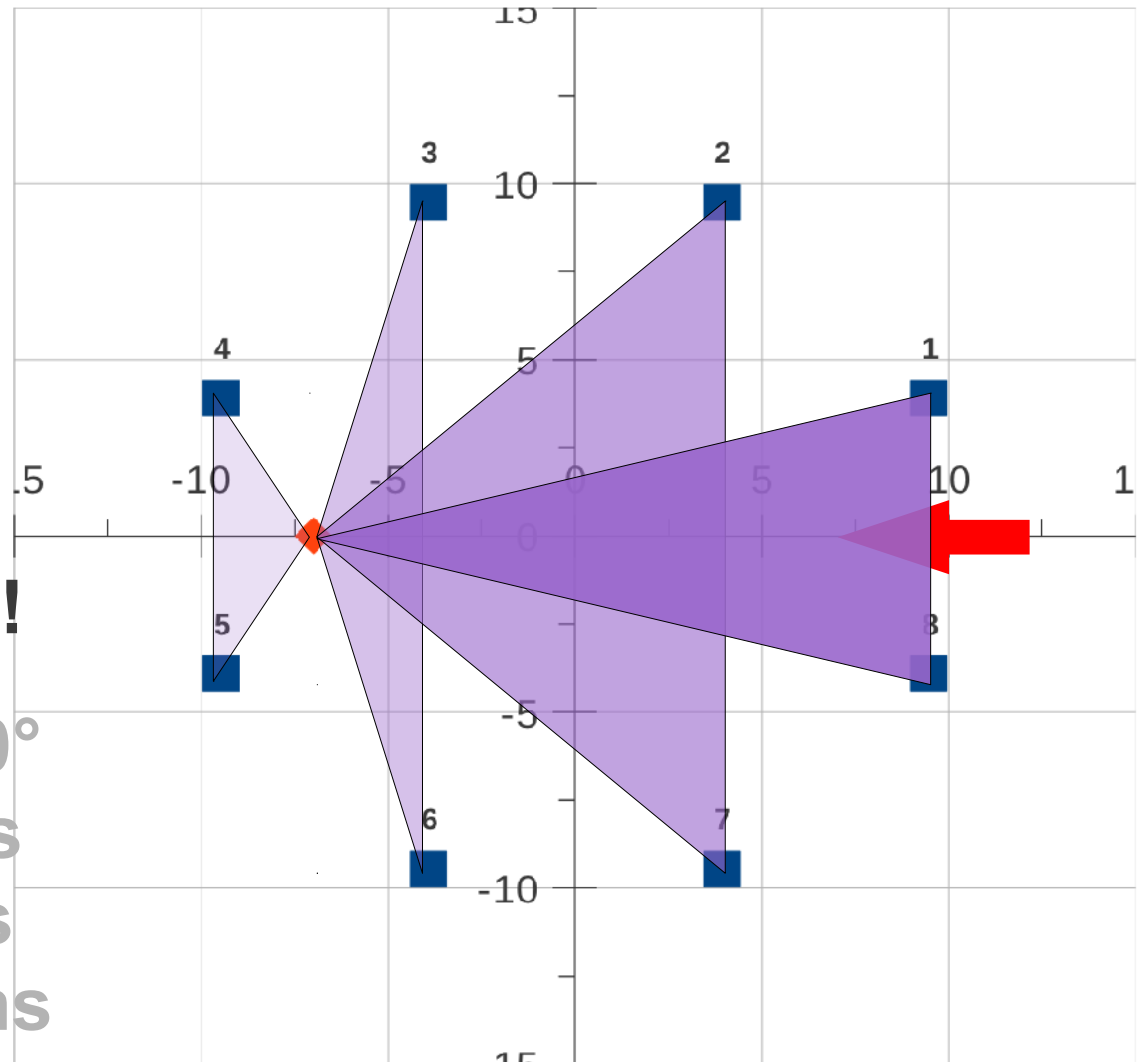


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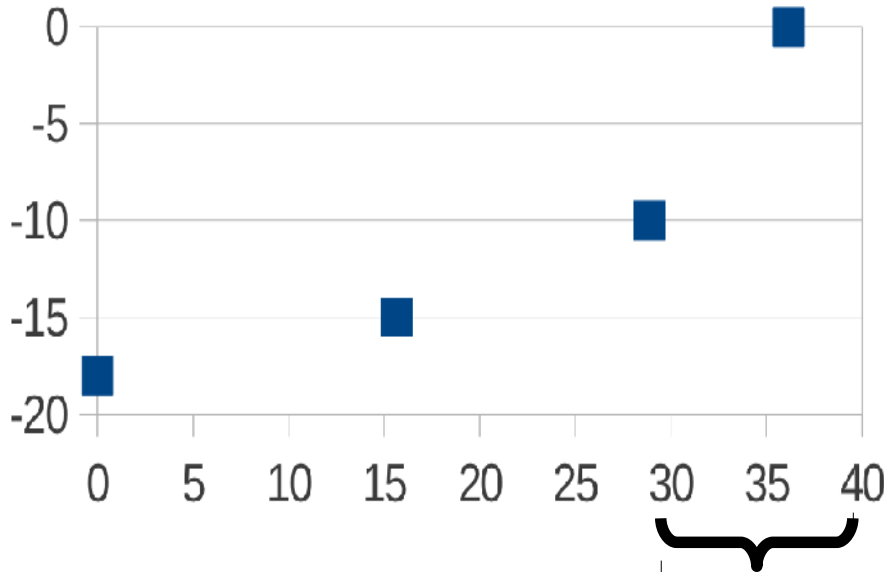


! ECHO threshold !

- 1st phantom source at 180°
- 2nd p.s. at 0° , +3dB, +16ms
- 3rd p.s. at 0° , +8dB, +29ms
- 4th p.s. at 0° , +18dB, +36ms

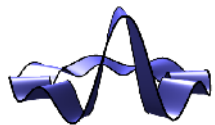
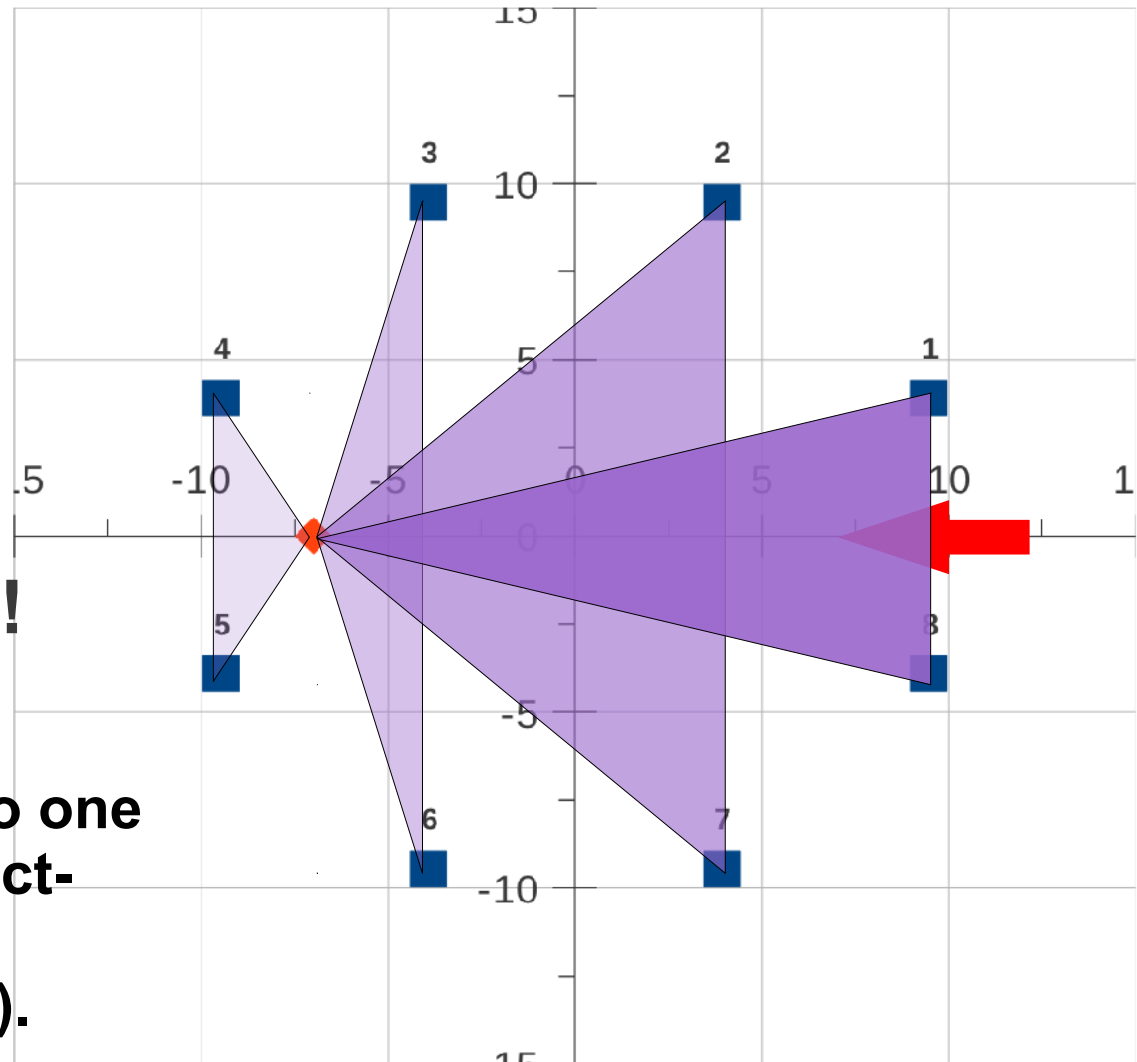


Theoretical considerations:



! ECHO threshold !

**We did not perceive any echoes.
The sonic events were fused into one
auditory event (→ listening expect-
ation on large system is totally
different from studio monitoring).**



Conclusion:

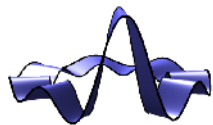
It works very well, although we don't quite understand why.

Backward masking?

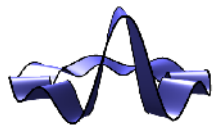
Reduced precedence effect when confounded by multiple sonic events with rising level?

Trading effect of echo threshold vs. number of sonic events?

Further rigorous listening tests are necessary.



**In practice, we feel
confident that a 3rd-order
Ambisonic system will
deliver good results for at
least 600 people
(300m² or r~10m).**



Acknowledgements:

Thanks to:

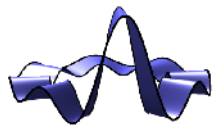
Adamson Europe
for the speaker systems
<http://adamsonsystems.com>



Cobra Sound Hamburg
for the space
<http://cobrasound.de>



The Linux Audio community
for the software.



Thank you.

I'm looking forward to your questions.

The paper is available at

http://stackingdwarves.net/public_stuff/linux_audio/ambisonic_symposium_2011/AmbiSym2011-Nettingsmeier-Dohrmann_Large-scale_HOA_Systems.pdf

