防災・減災情報を効果的に伝送する メッシュネットワーク型インテリジェント 拡声システムの研究開発

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Information flow on warning system

 \rightarrow multiple paths to deliver disaster messages.







In the Great East Japan Earthquake, among residents who could obtain evacuation announcements in the affected prefectures, 45% of them obtained it from outdoor loudspeakers.

However, among those who obtained evacuation announcements from outdoor loudspeakers, only 56% of them answered they could clearly hear the announcements.



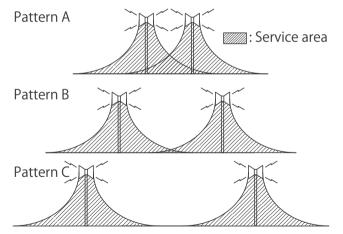


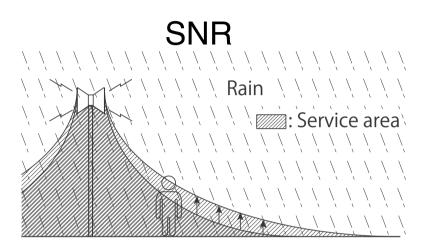
- <u>Problems</u> : Residents in a house. <u>Difficulty on hearing in a field.</u>
- Reasons : Overlapping of sounds from other public address system or reflected sounds by building, mountain, woods, and so on.

What functions are expected?

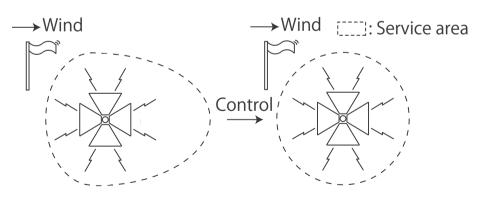
- Timing control
- Sound pressure level control
- Directivity control

Sound overlapping

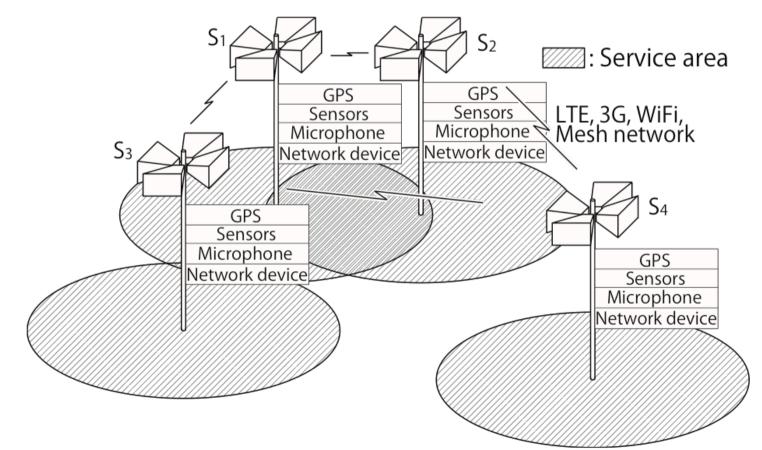




Wind / Temperature

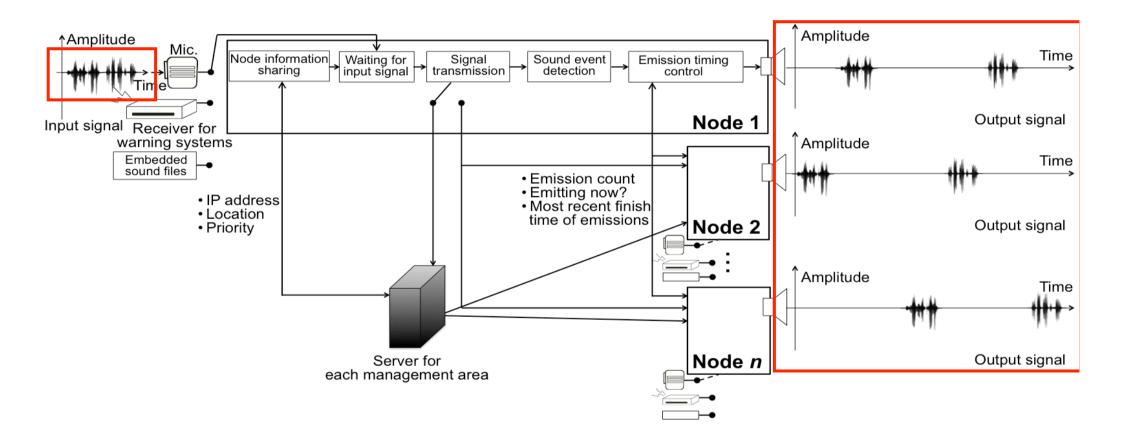


• Timing control is implemented to improve signal to interference ration at overlapped area.



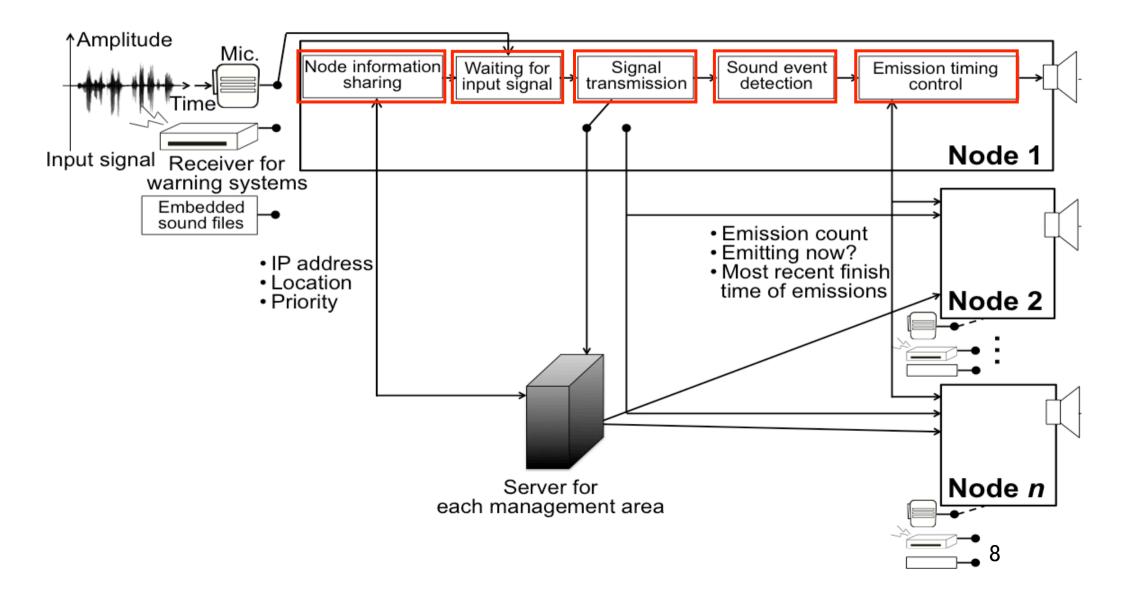
Emission timing controller

Emission timing control



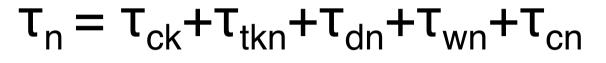
Emission timing controller

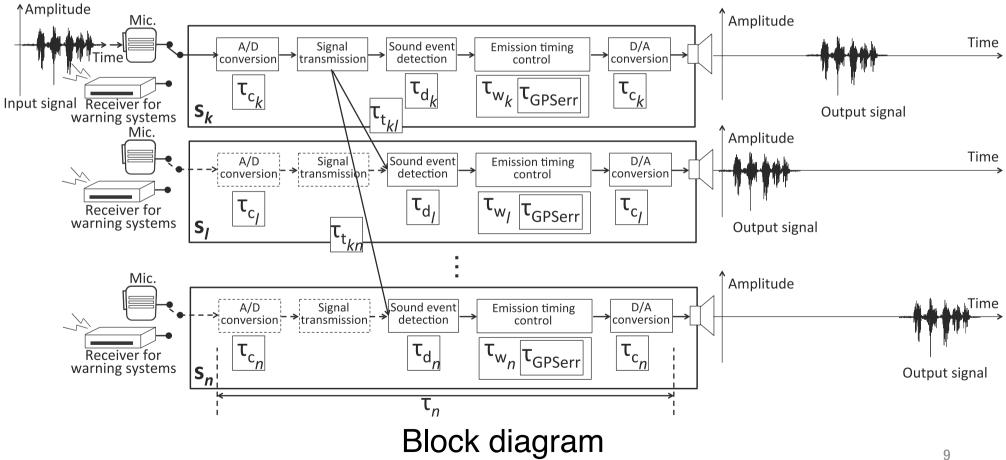
Emission timing control



Processing time

total delay from input to output





Errors on each block

total delay $T_n = T_{ck} + T_{tkn} + T_{dn} + T_{wn} + T_{cn}$

 τ_{ck} , τ_{cn} : A/D, D/A conversion τ_{tkn} : transmission time 1 frame data(bits) / throughput(bps) depends on network. e.g. PCM 16 bits, 1 frame = 1024 samples, LTE 2 Mbps < 10 ms τ_{dn} : sound event detection e.g. 10 ms(ITU G7.29 AnnexB) τ_{wn} : emission timing control

Intelligent public address system

The detailed system and its performance is described in

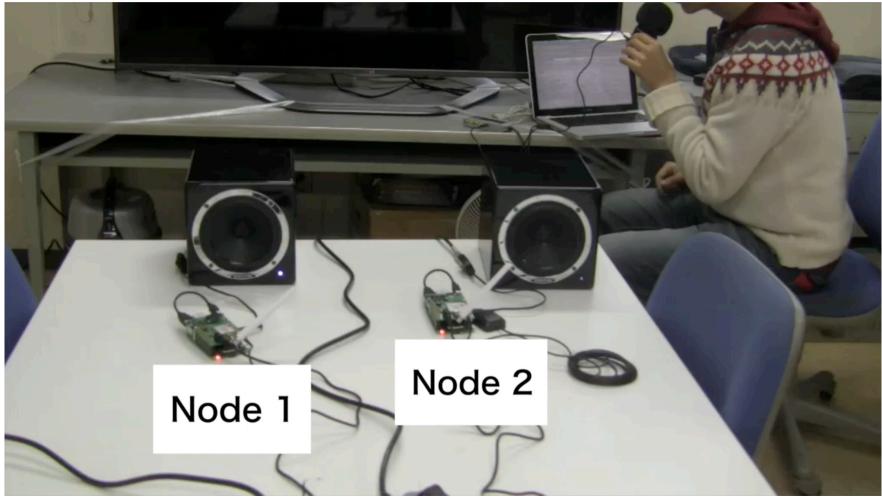
Emission timing control method for improving signal to interference ratio on public address system

Taira Onoguchi, , Dan Murakami, Yoshifumi Chisaki

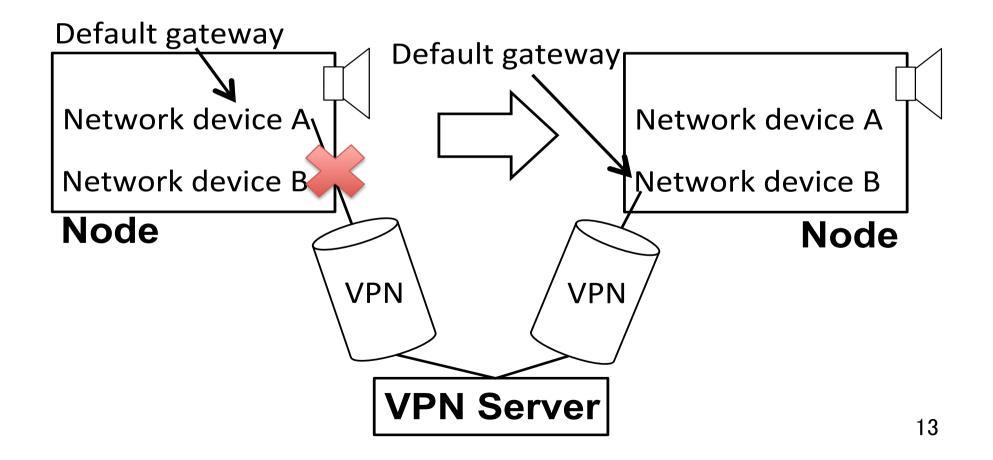
<u>Applied Acoustics</u> <u>Volume 98, November 2015, Pages 70–78</u>

Demonstration

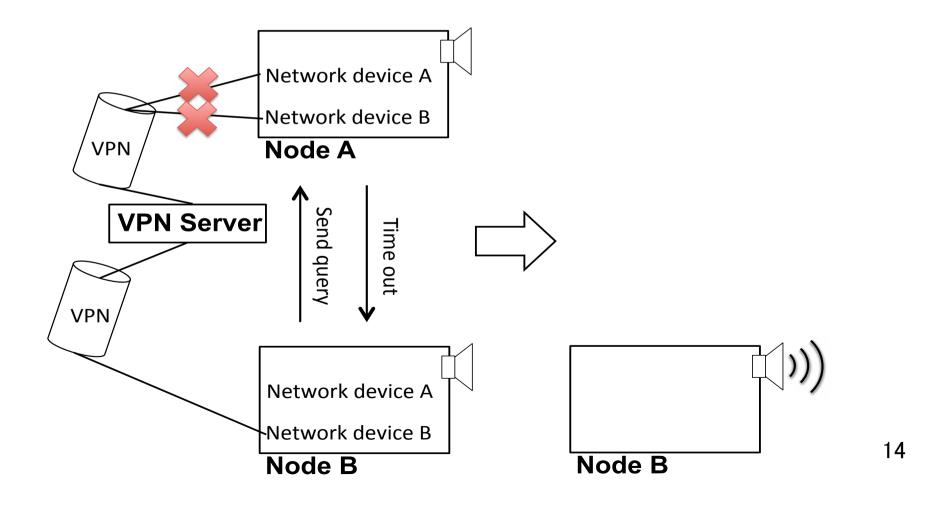
(Residents in Kurokami area, move to higher place in Japanese.)



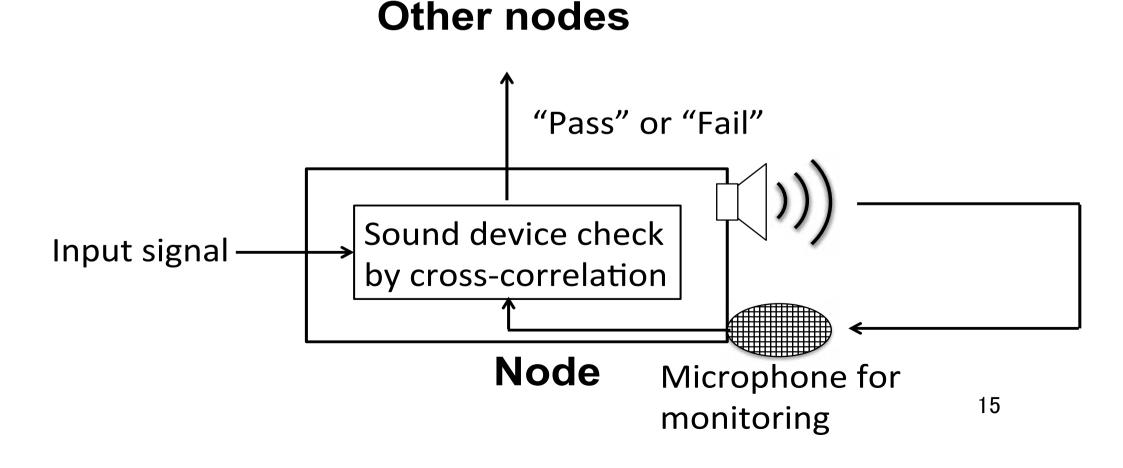
- Automatic network selector
 - Nodes and a server automatically change their default gateway to select the active and widest bandwidth network.



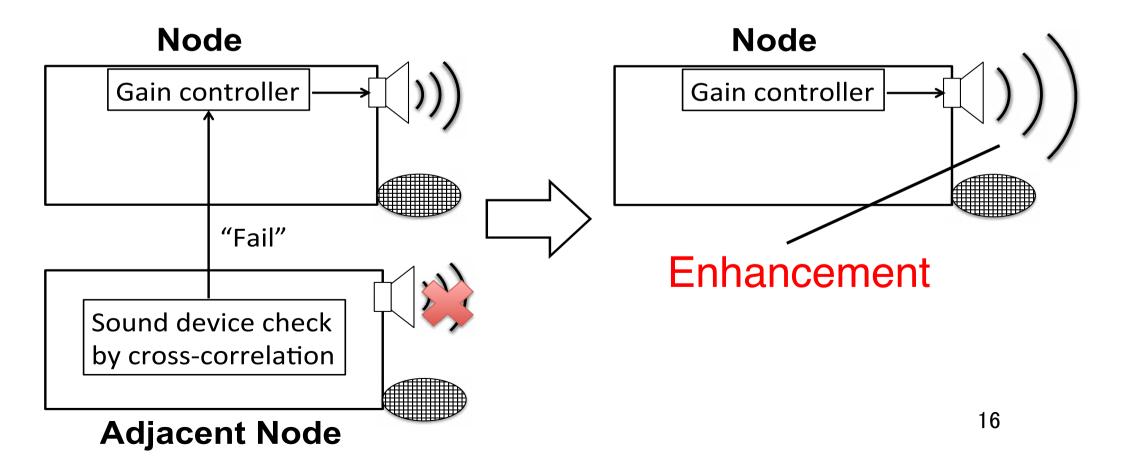
- Fail safe structure for network disorder
 - If all of network devices of a node are down, other nodes determine emission timing without considering that node.



- Monitoring own emission and environmental sound
 - Monitoring own daily announcements to check the status of sound output.



- Emission sound level enhancement
 - Status of sound output is shared with other nodes.
 - If sound output of the adjacent node is down, the node increases its emission sound level.



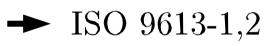
Introduction

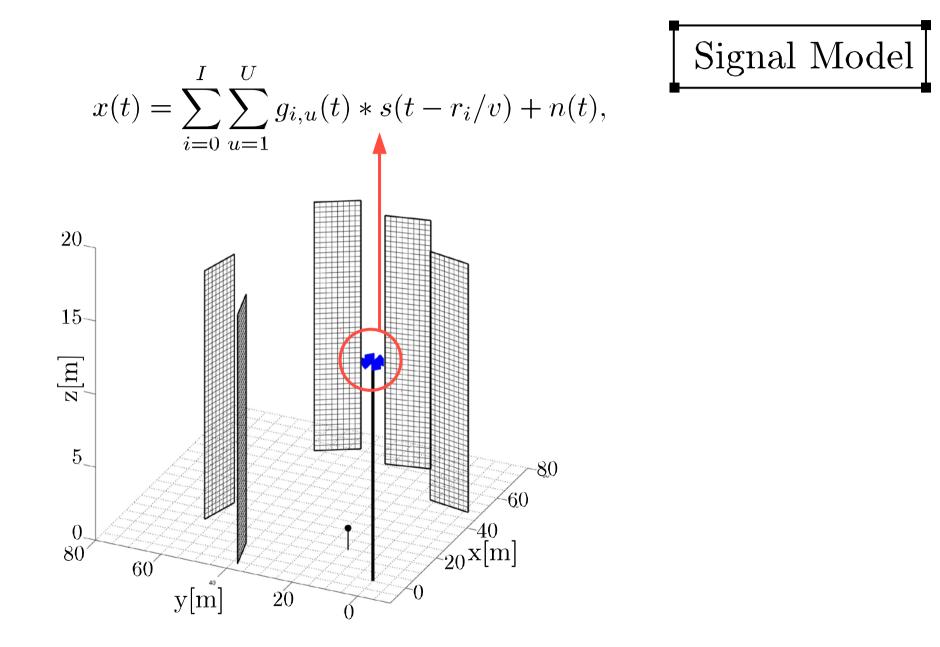
- Estimation of time difference of arrivals between direct sound and reflected sounds using time-frequency information of a single-channel signal. *Acoust. Sci. & Tech.* (in press).

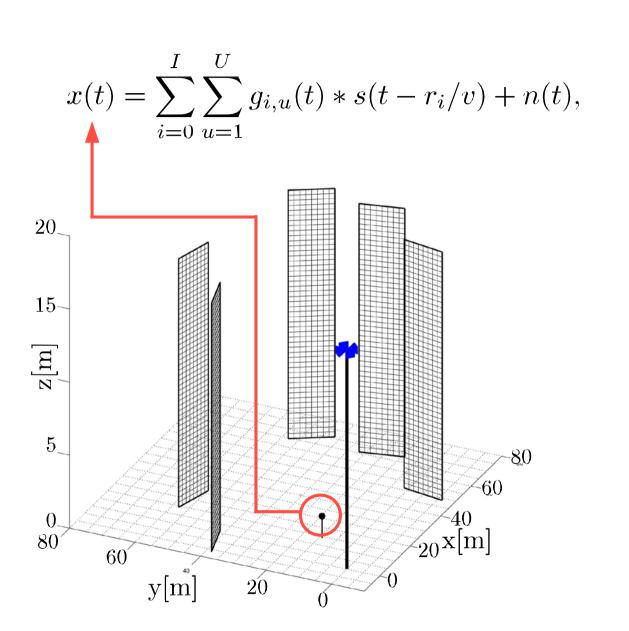
- did not take into account the attenuations due to
 - atmospheric absorption
 - size of obstacles

Previous study

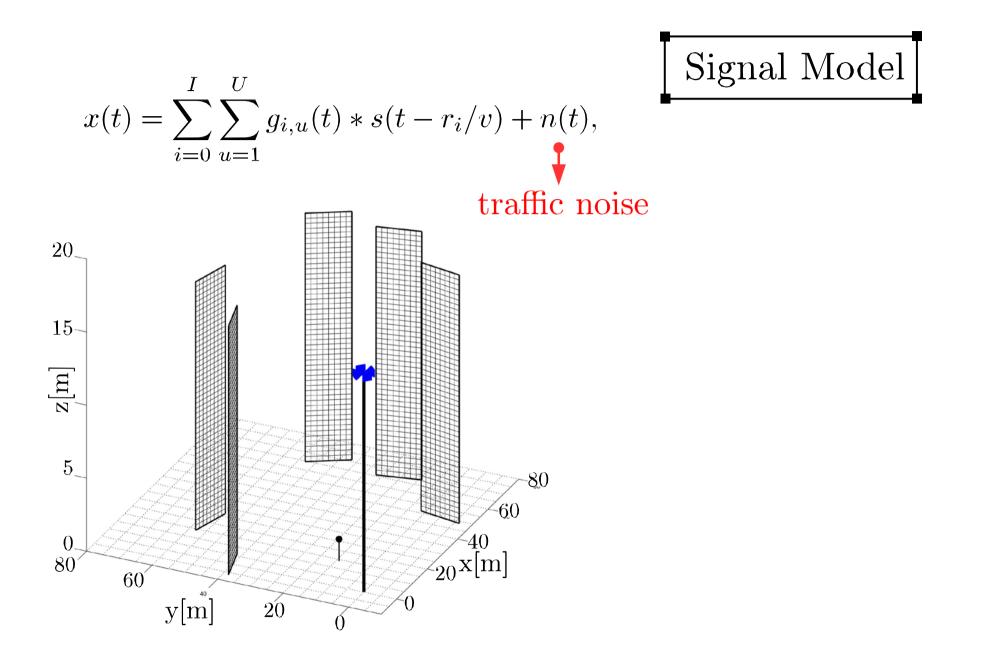
- This paper examines the previous method
 - atmospheric absorption
 - reflections from obstacles
- Numerical tests of TDOAs estimation
 - to evaluate the performance

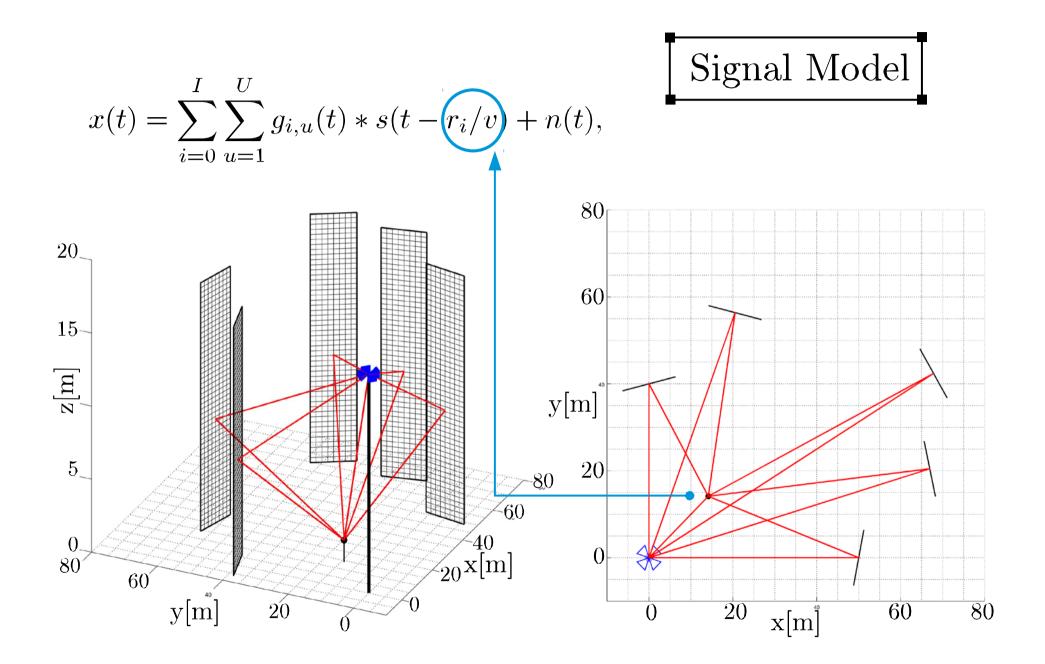


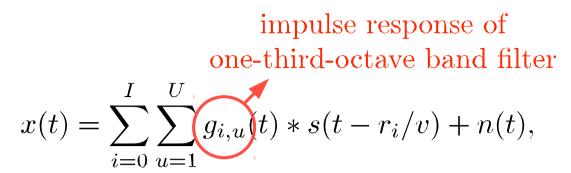




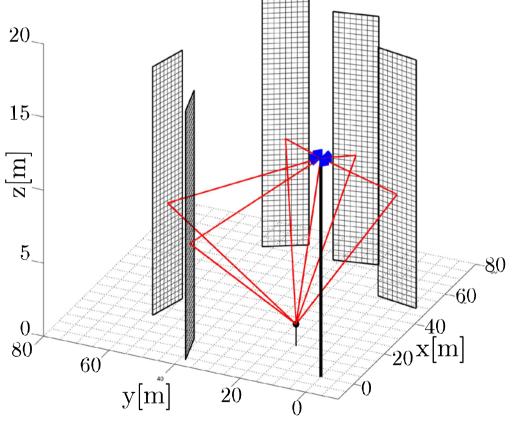




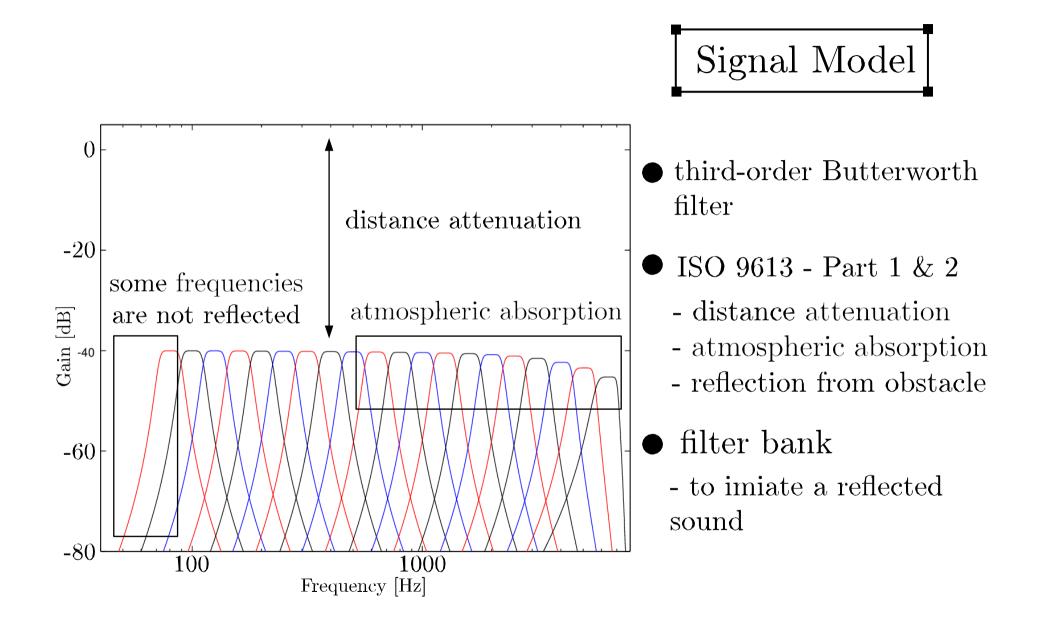


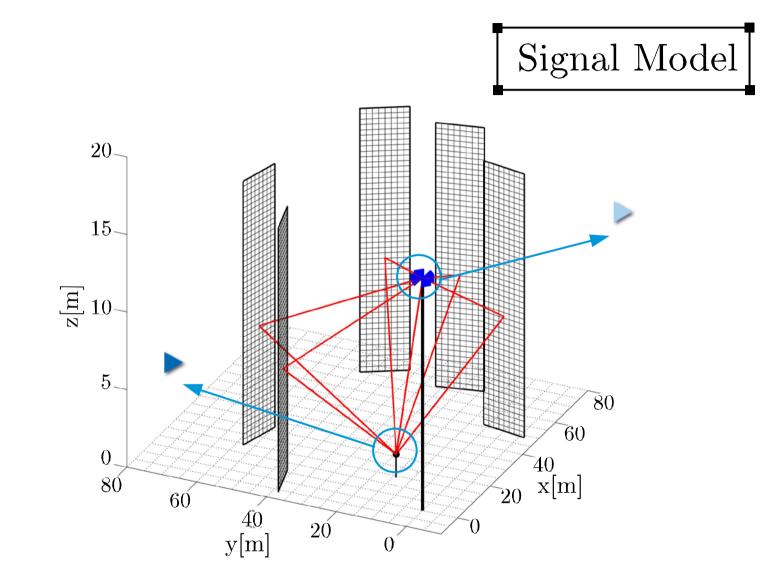


Signal Model



- distance attenuation
- \bullet atmospheric absorption
 - air temperature, $15^{\circ}\mathrm{C}$
 - relative humidity 73%
 - air pressure 101.325 kPa
- reflection from obstacle
 - sound reflection coefficient is assumed to be 0.8
 - the size of obstacle is large enough compared to wavelength







(1) TDOAs estimation

- to extract peaks indicating TDOAs from a single channel signal

- (2) Peak identification process
 - to provide a better visualization
 - to easily identify and obtain peaks

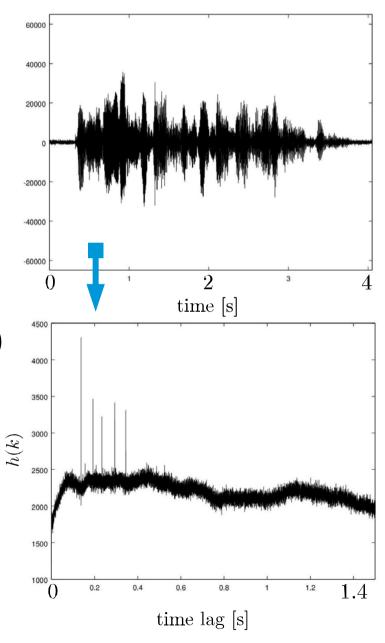
(1) <u>TDOAs estimation</u>

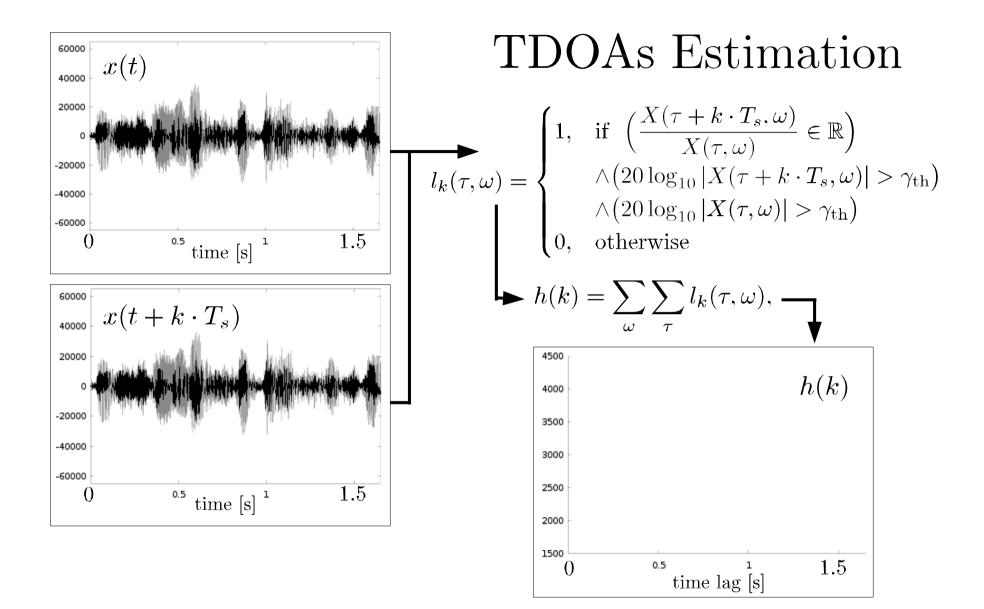
• to find TDOAs, we introduce and define

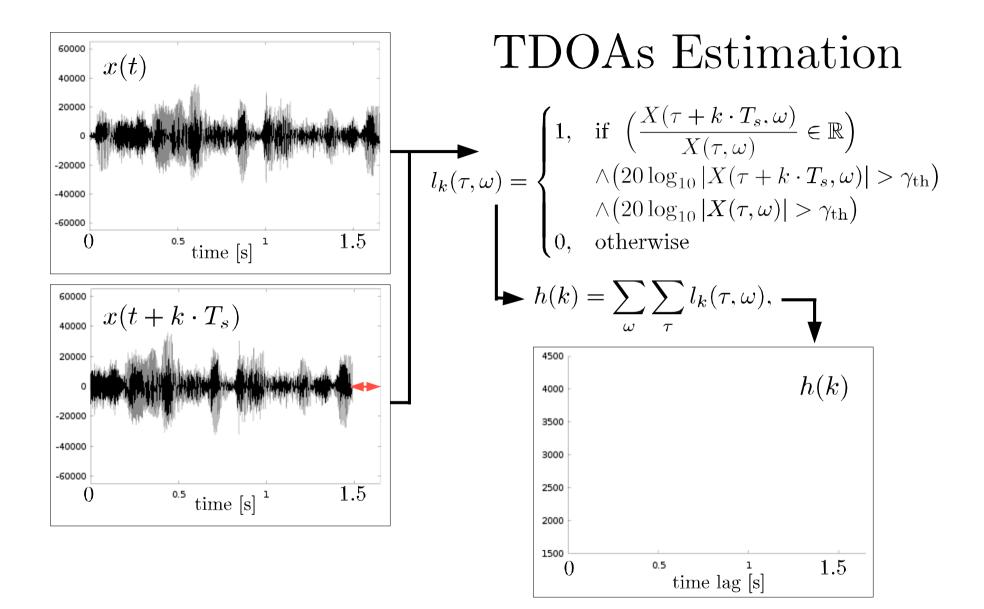
$$h(k) = \sum_{\omega} \sum_{\tau} l_k(\tau, \omega),$$

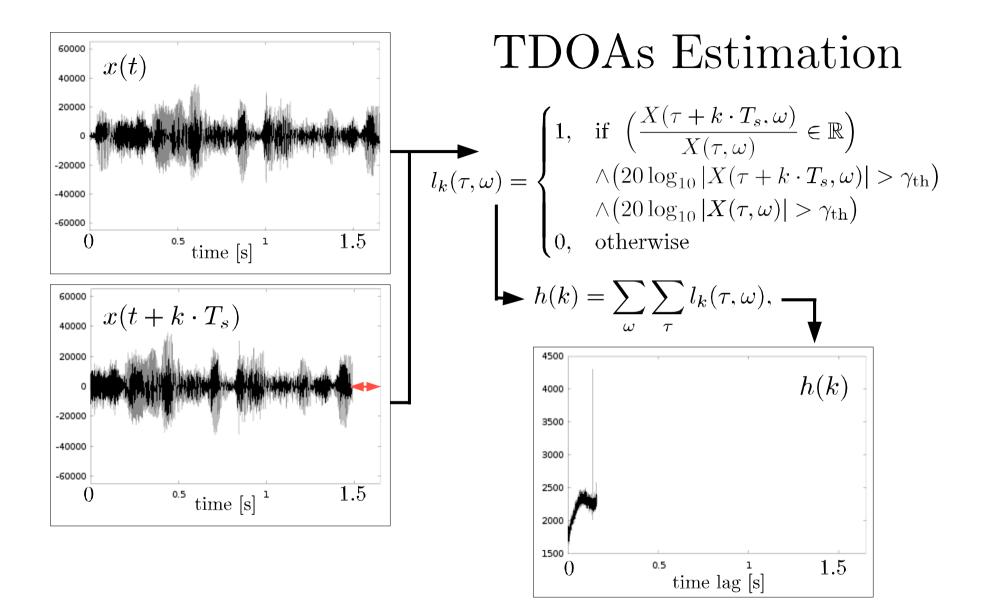
$$\bullet \\ l_k(\tau, \omega) = \begin{cases} 1, & \text{if } \left(\frac{X(\tau + k \cdot T_s, \omega)}{X(\tau, \omega)} \in \mathbb{R} \right) \\ & \wedge \left(20 \log_{10} |X(\tau + k \cdot T_s, \omega)| > \gamma_{\text{th}} \right) \\ & \wedge \left(20 \log_{10} |X(\tau, \omega)| > \gamma_{\text{th}} \right) \\ & 0, & \text{otherwise} \end{cases}$$

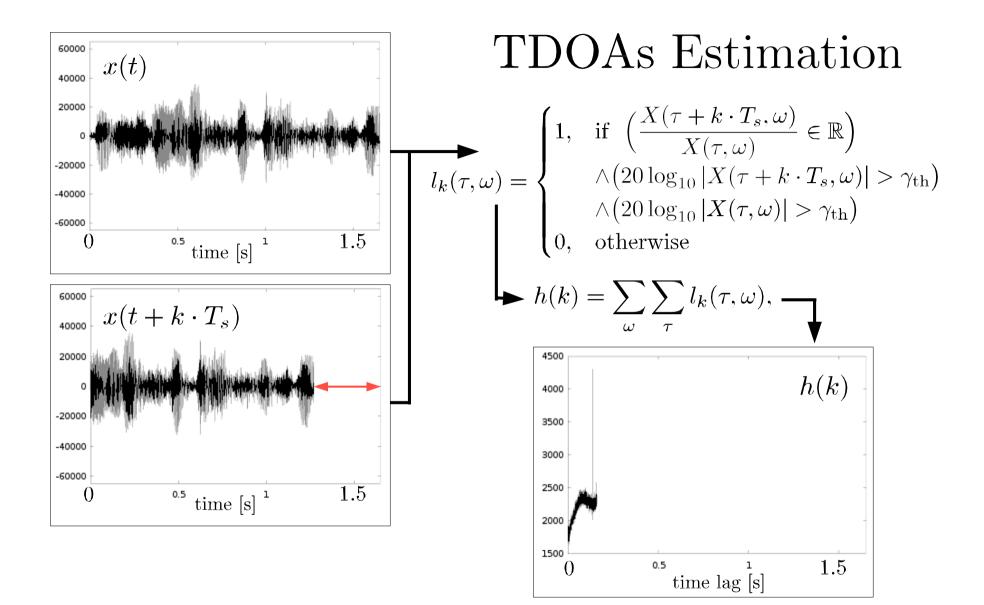
- $X(\tau + k \cdot T_s, \omega)$ is the STFT of $x(t + k \cdot T_s)$
- T_s is the sampling period
- $\gamma_{\rm th}$ is the background noise level in dB

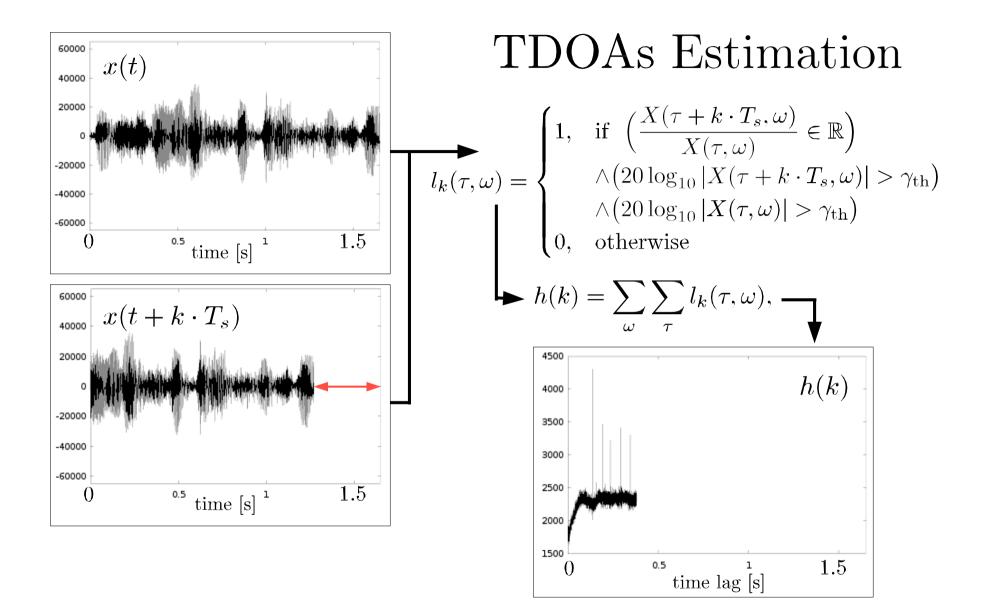


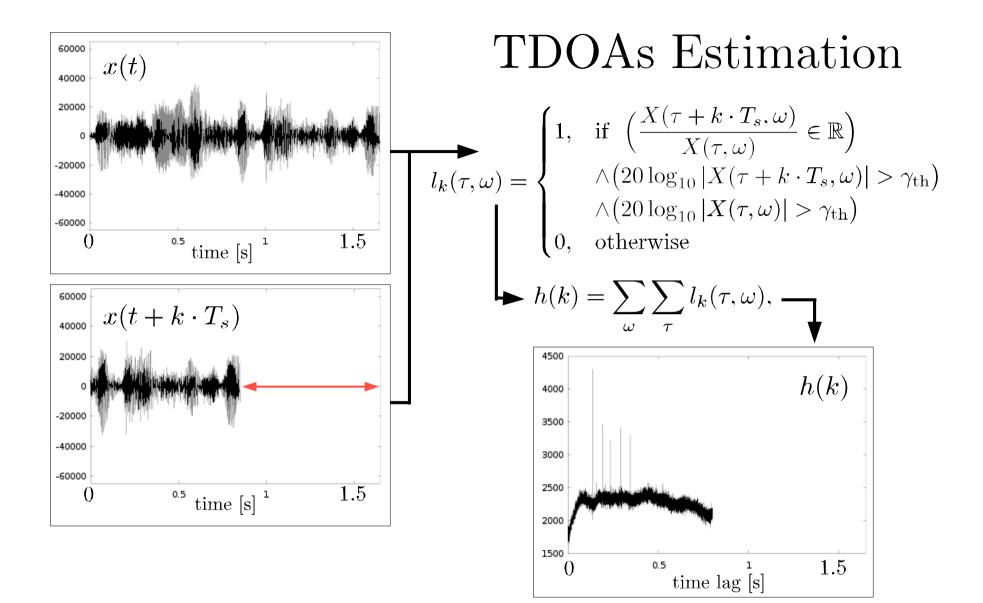










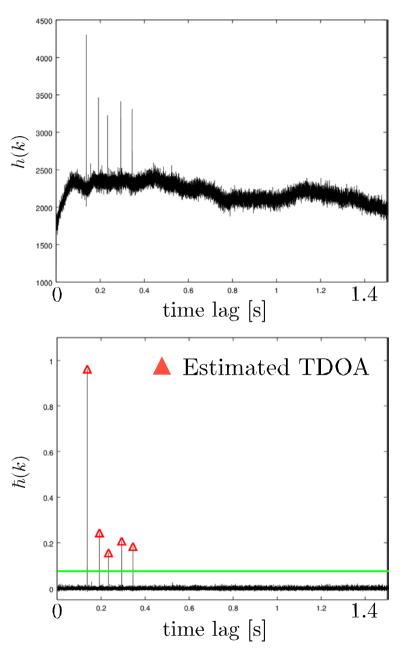


$(2) \underline{\text{Peak Identification}} \\ \underline{\text{Process}}$

- h(k) is divided into groups
- the data in each group are divided by the median value
- all groups are merged into the original size which is denoted as g(k)

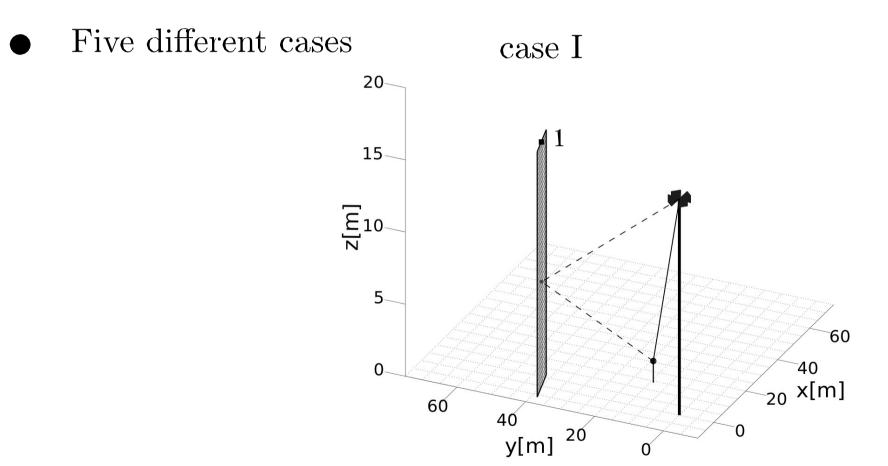
•
$$\hbar(k) = \left(\frac{g(k)}{\max(g(k))}\right)^{\beta} - \mathbf{E}\left[\left(\frac{g(k)}{\max(g(k))}\right)^{\beta}\right],$$

using a threshold to find peaks



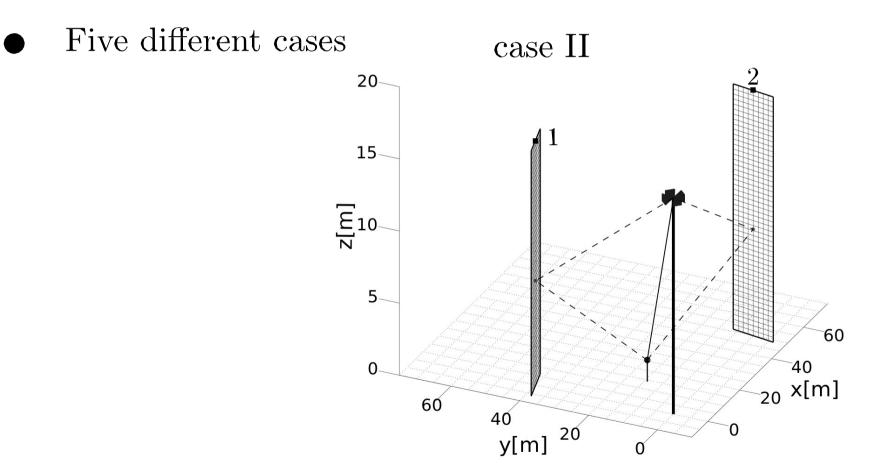
Result Discussion

For evaluation



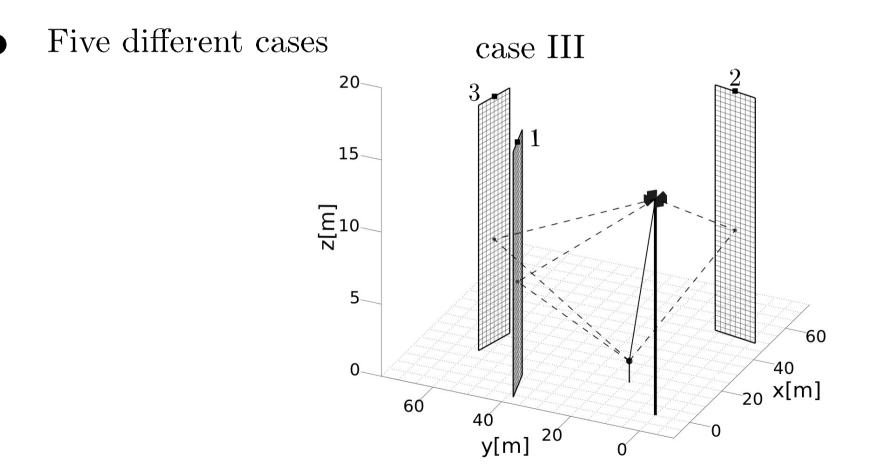
Result Discussion

For evaluation



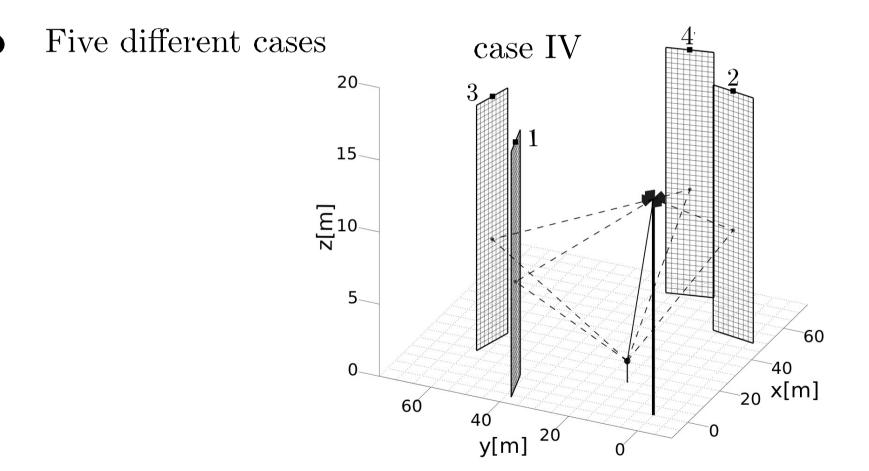
Result Discussion

For evaluation



Result Discussion

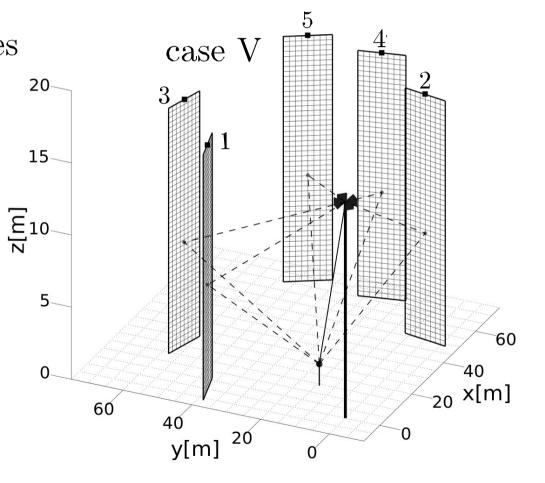
For evaluation



Result Discussion

For evaluation

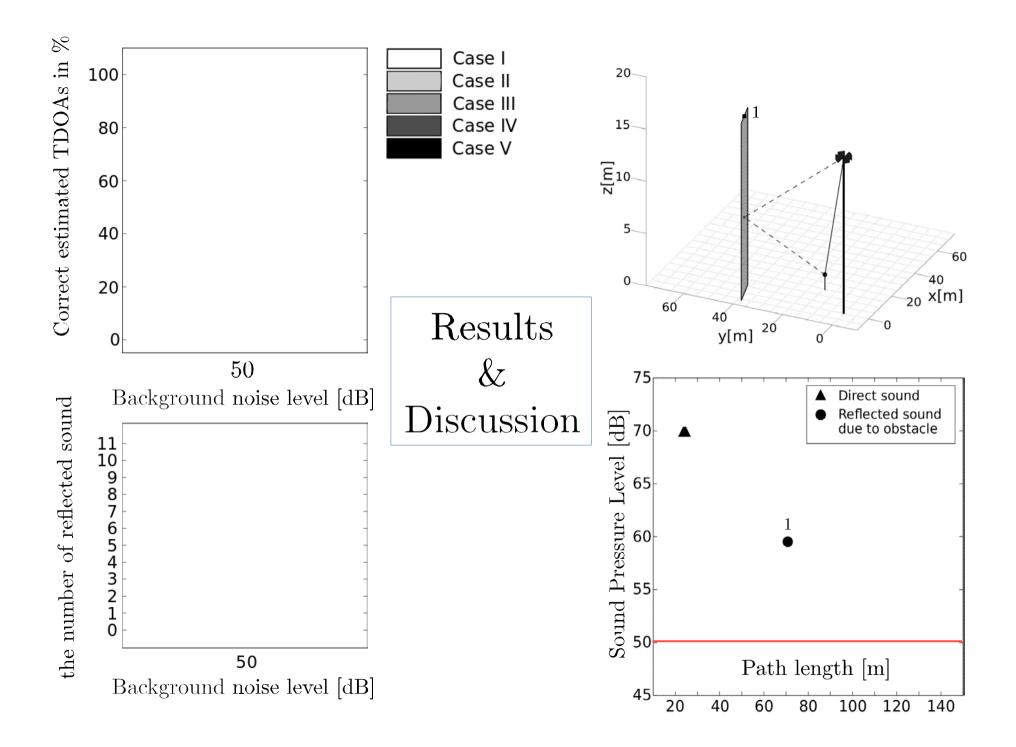
• Five different cases

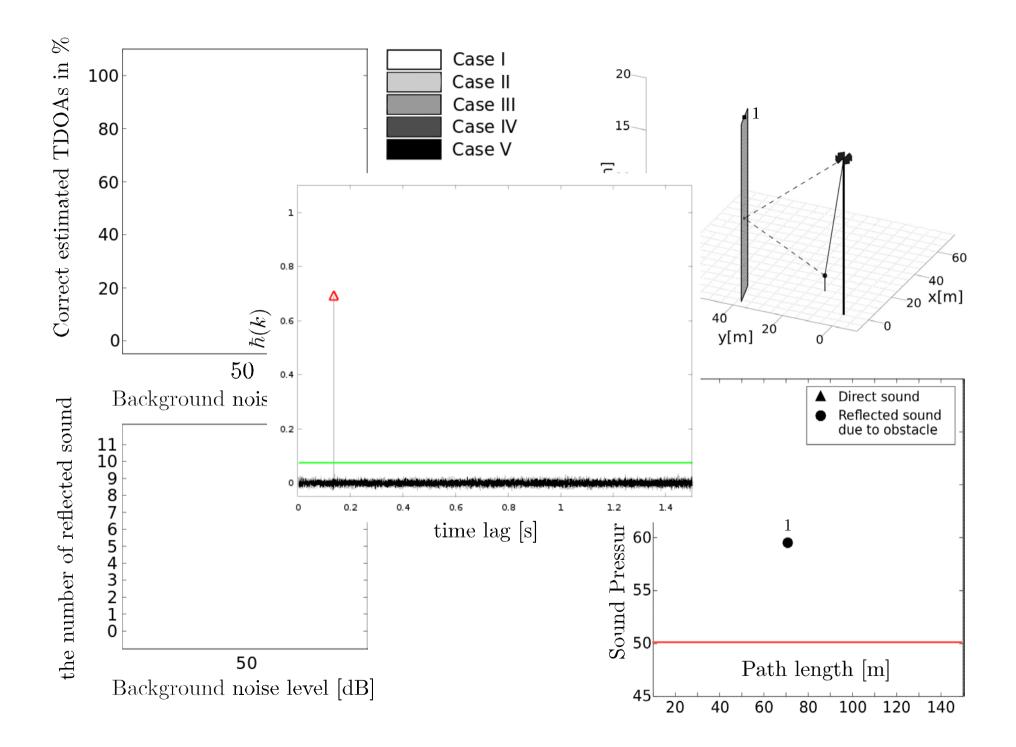


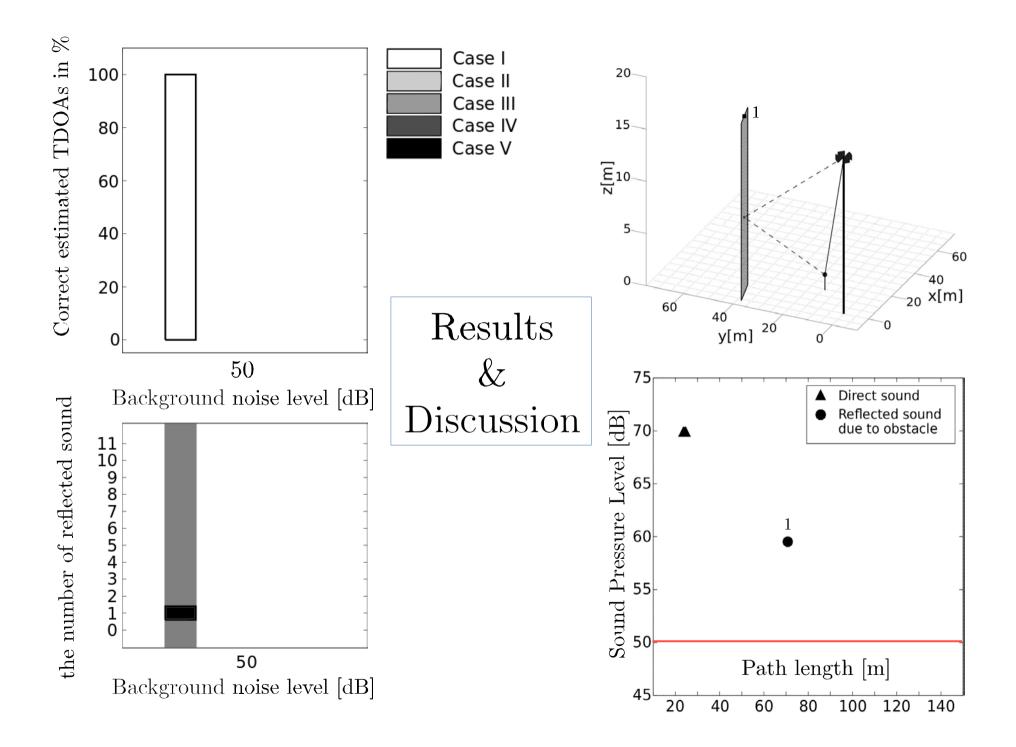
Result Discussion

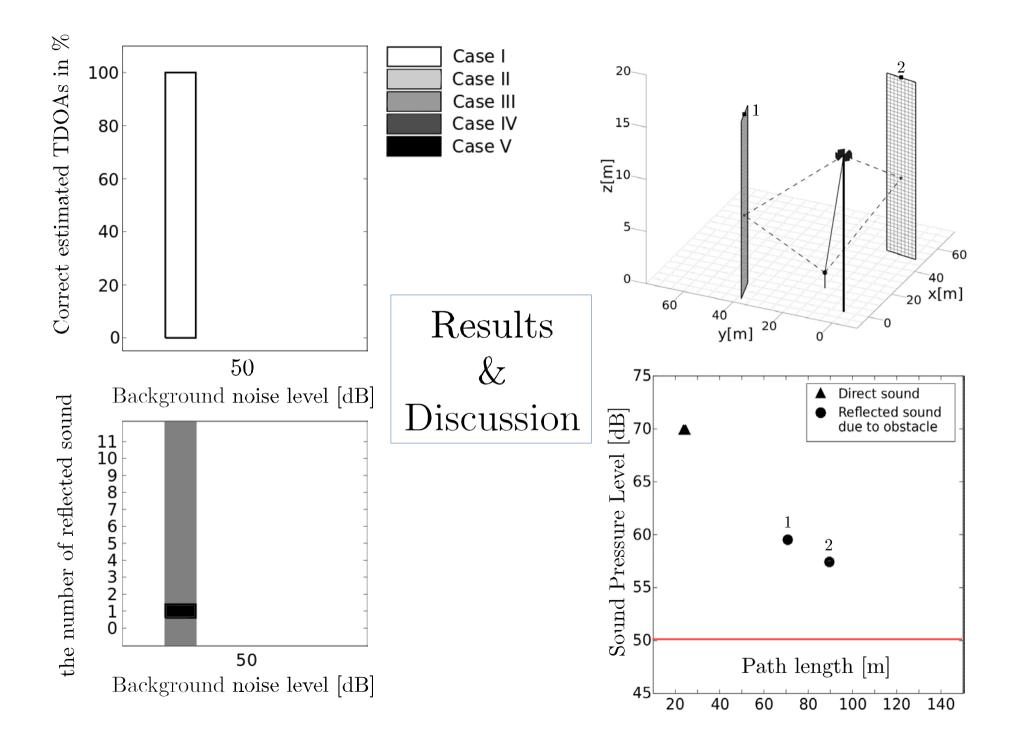
For evaluation

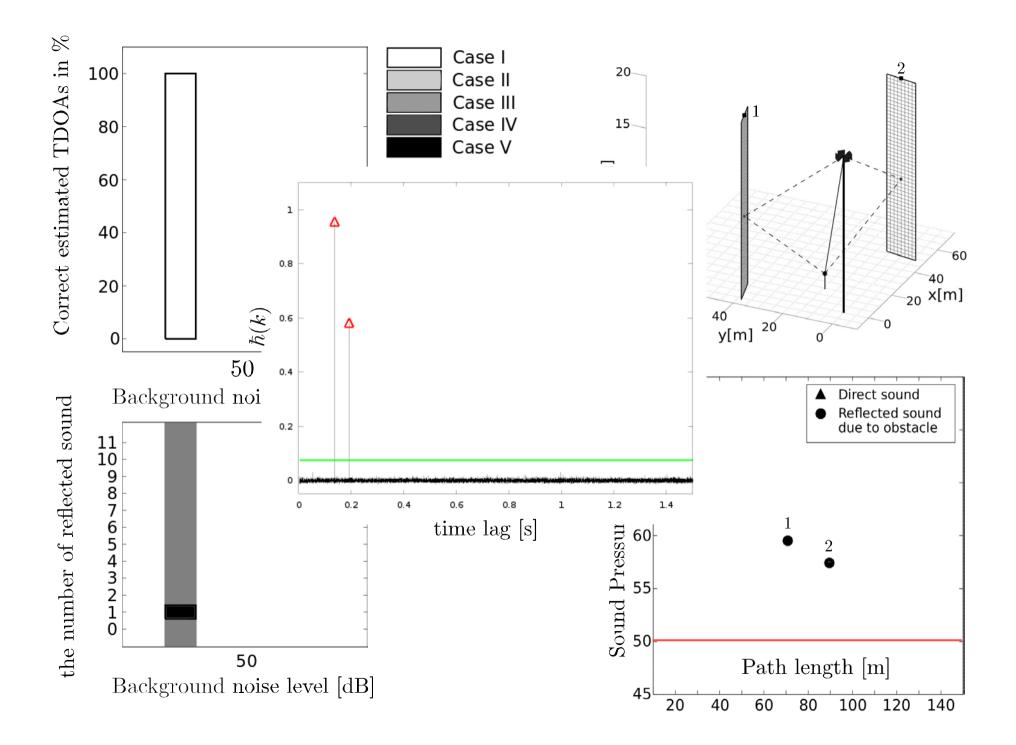
- Five different cases
- Evaluation criteria
 - the percentage of correct estimated TDOAs
 - histogram of the number of reflected sound

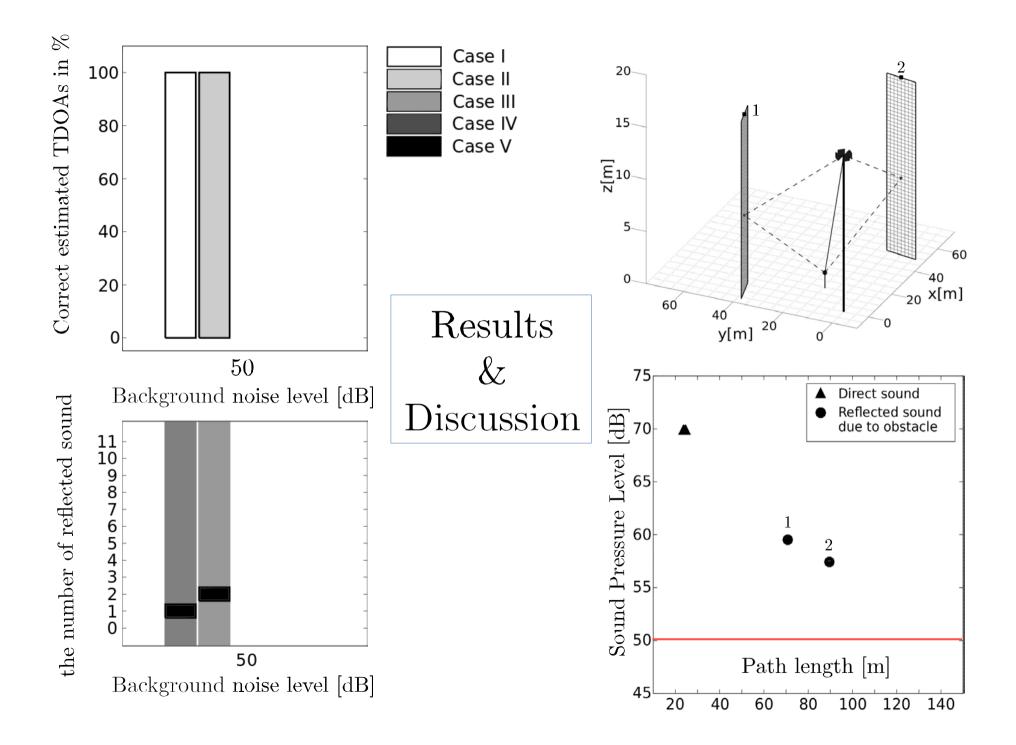


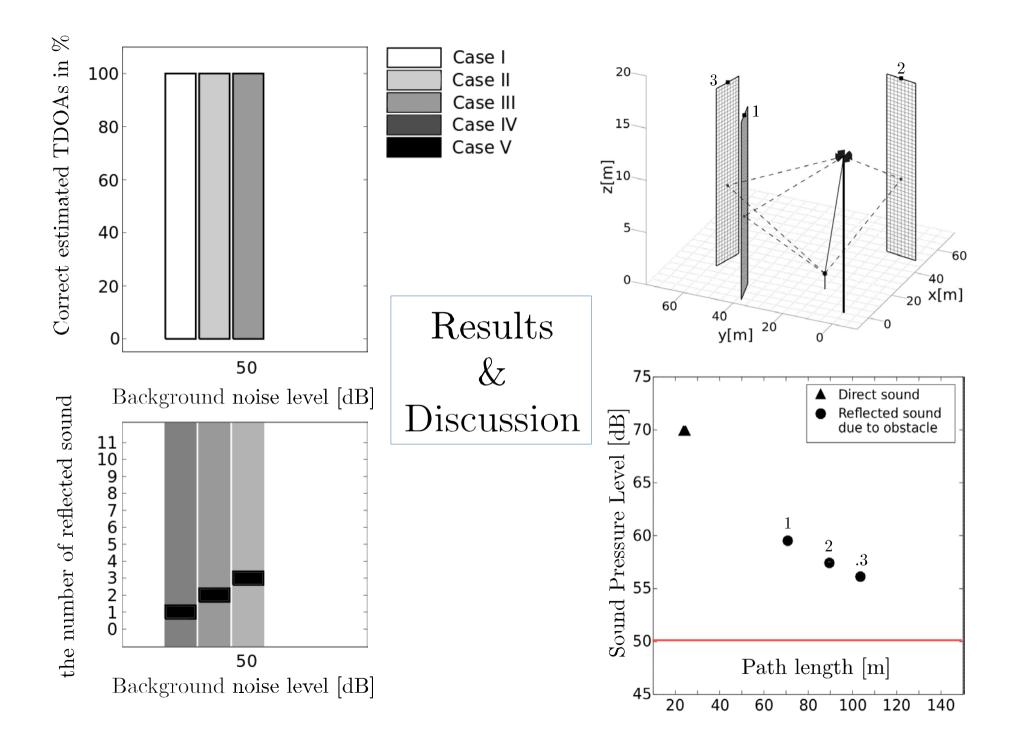


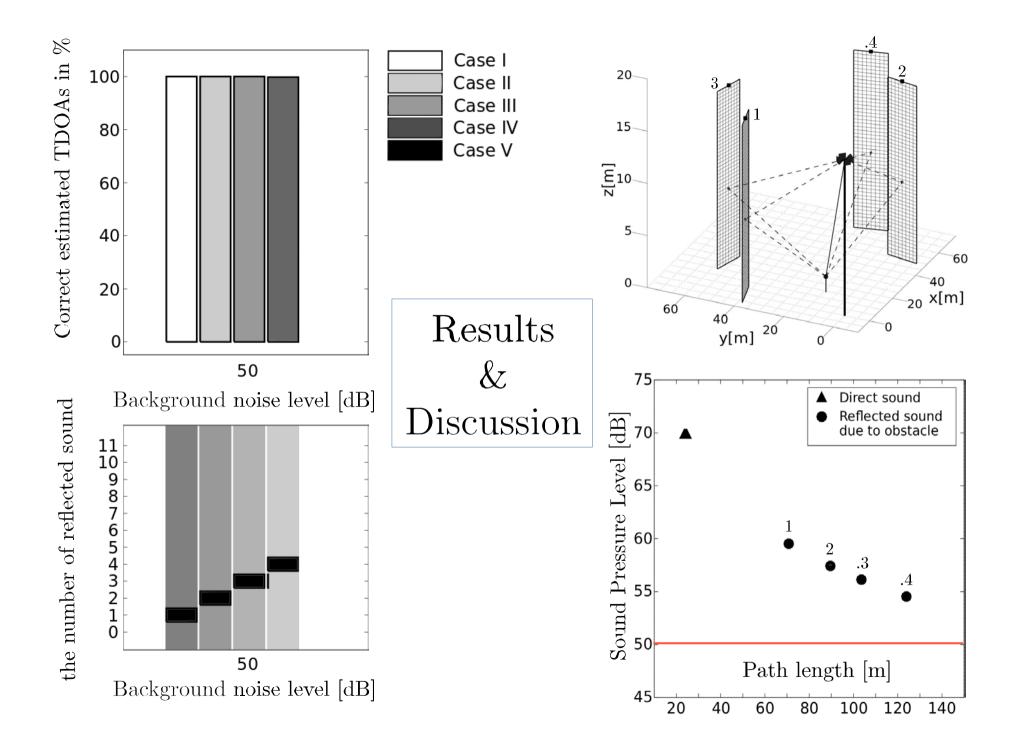


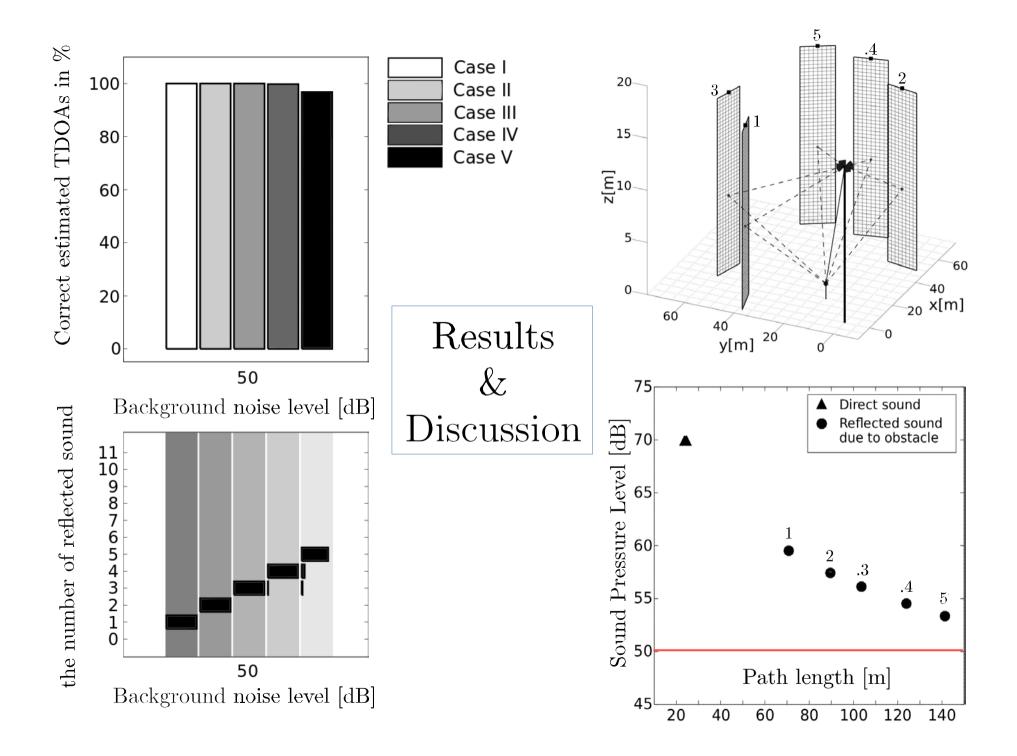


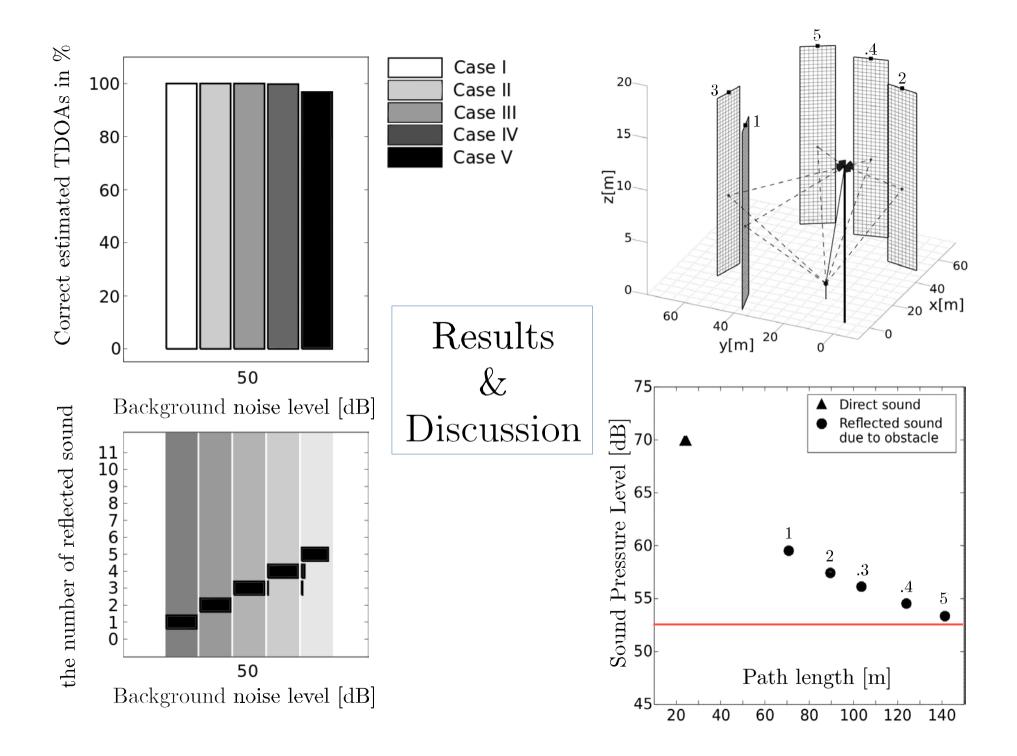


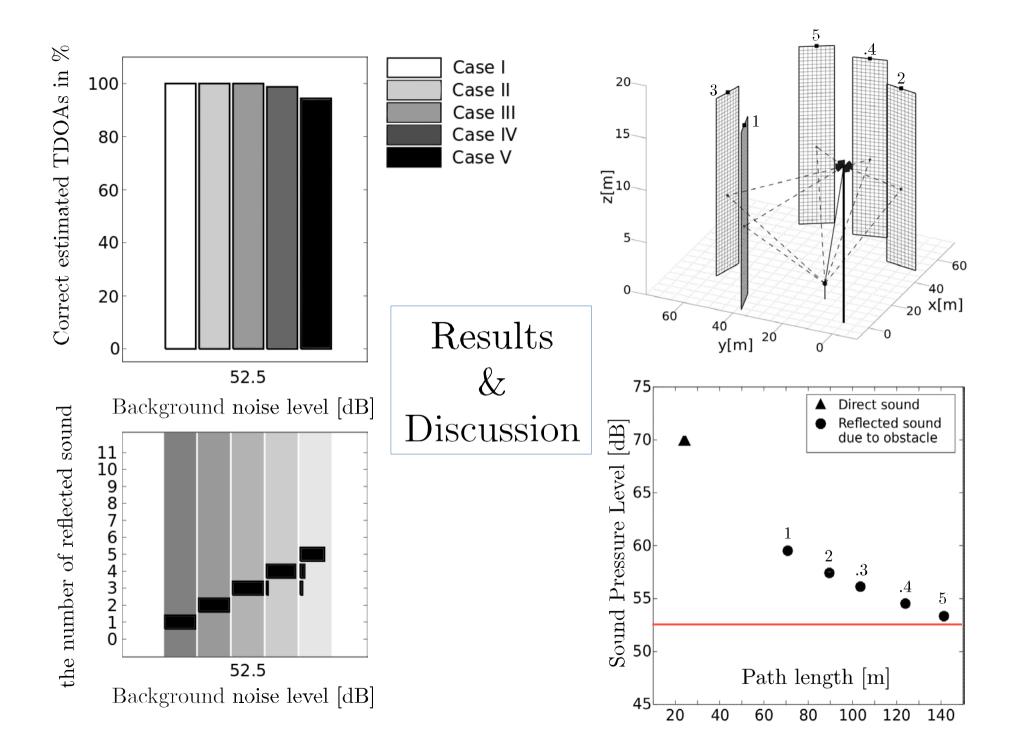


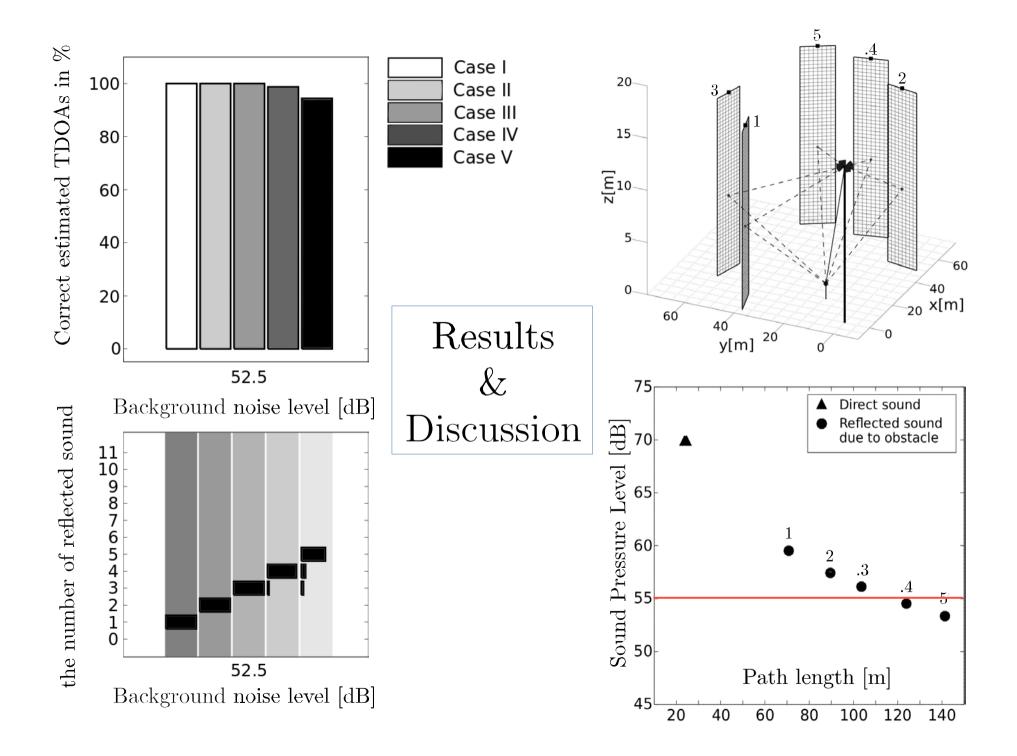


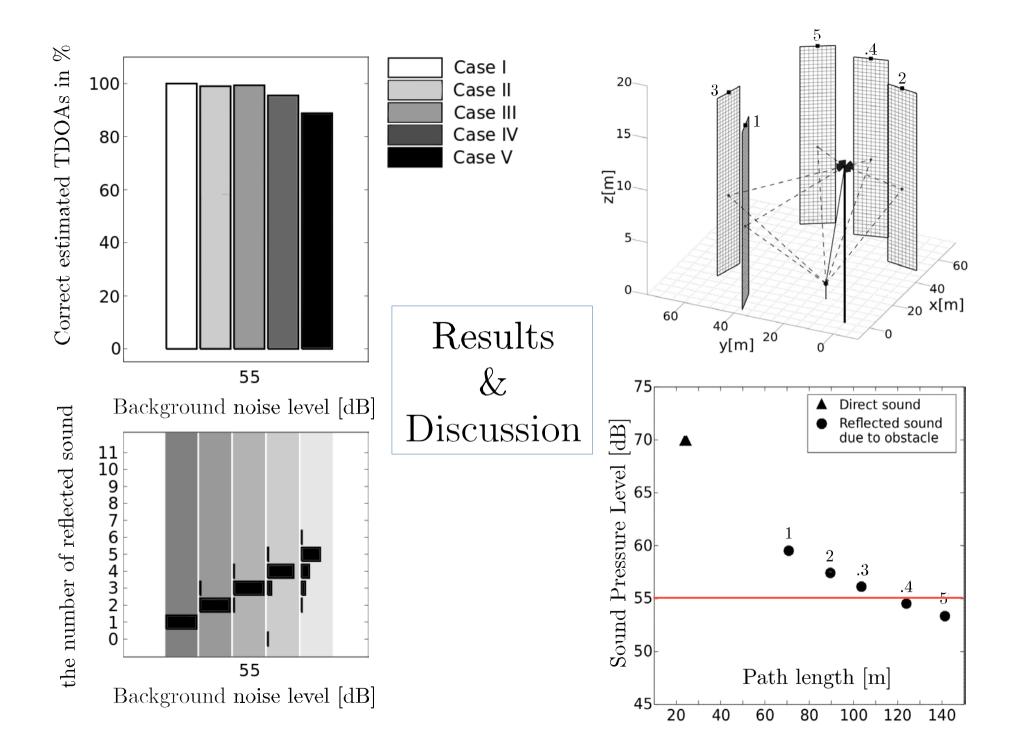


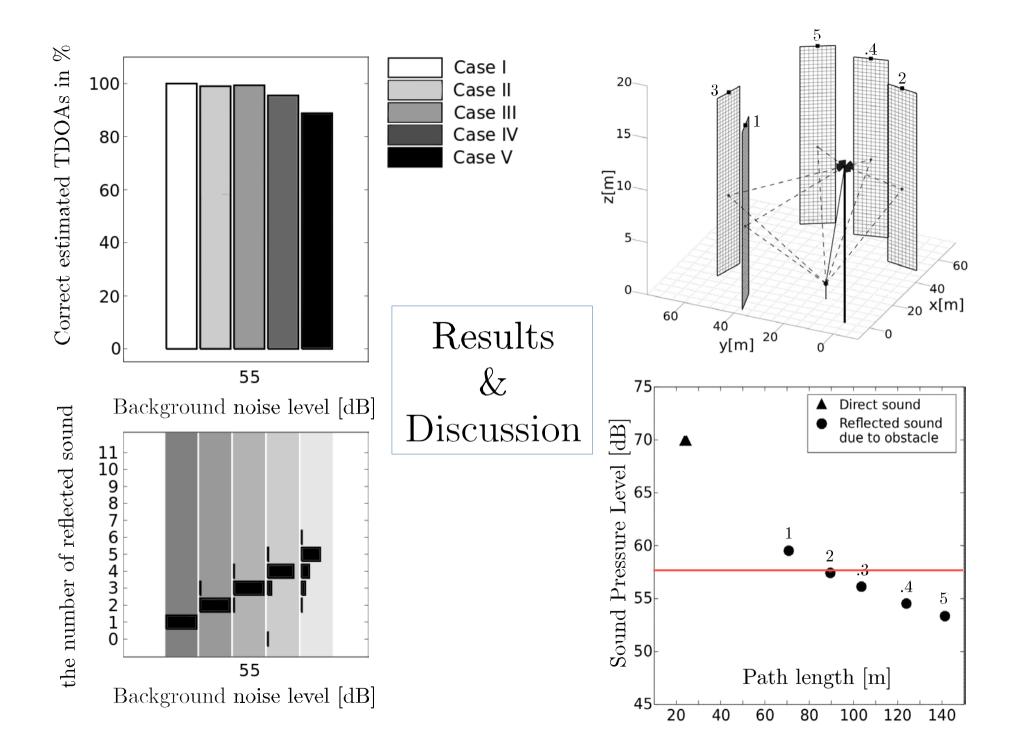


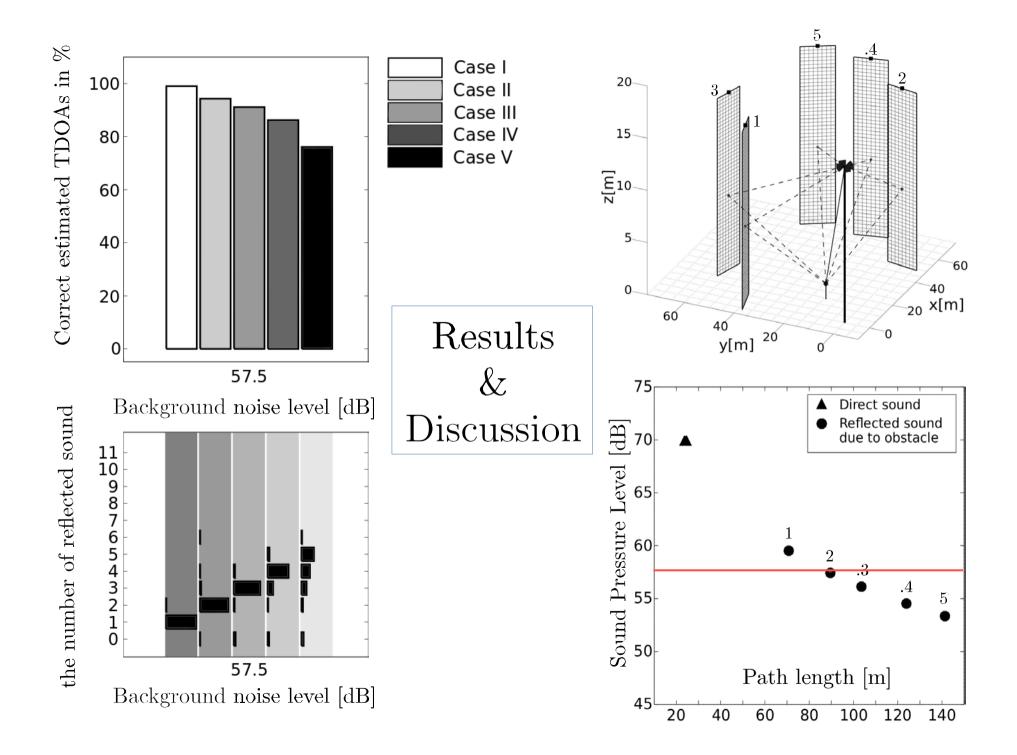


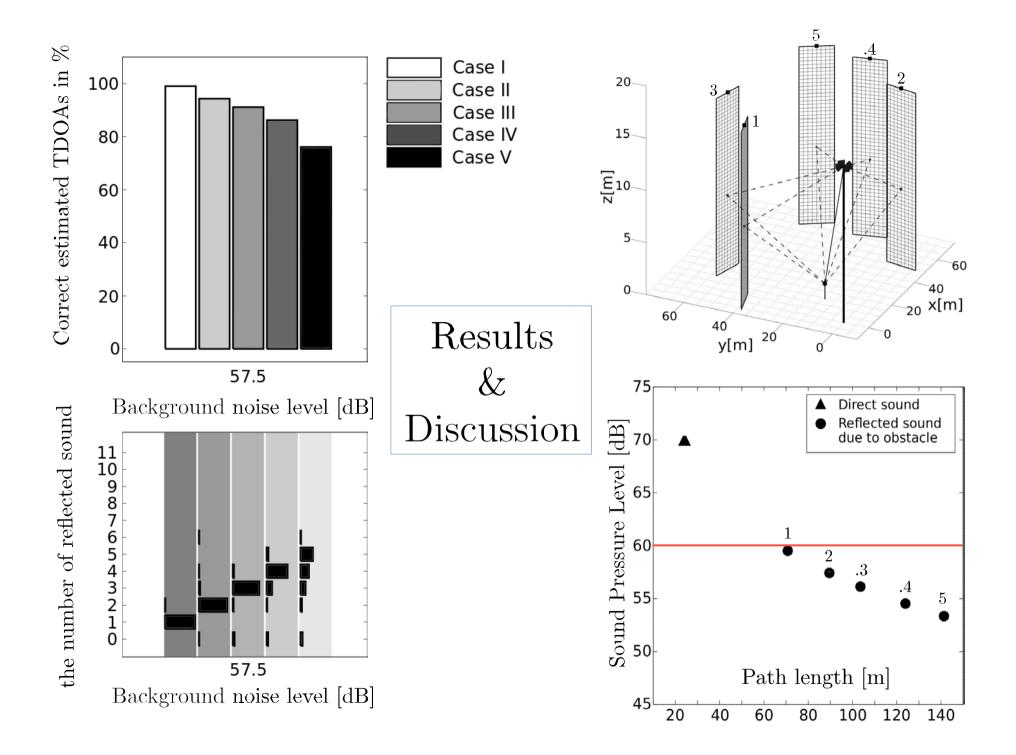


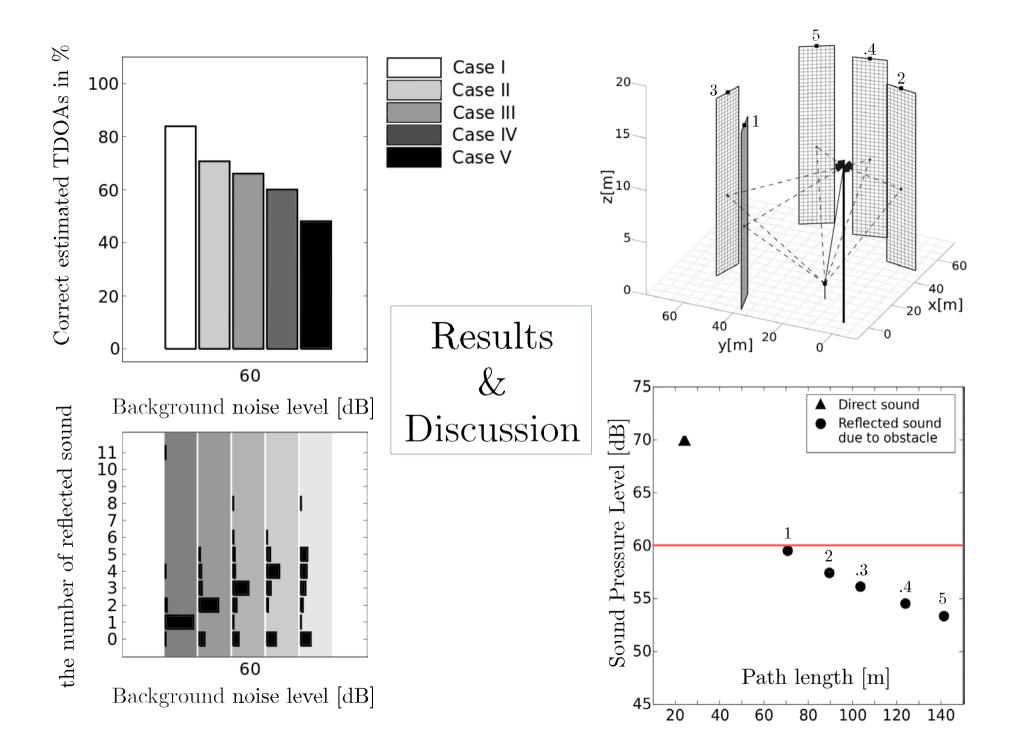


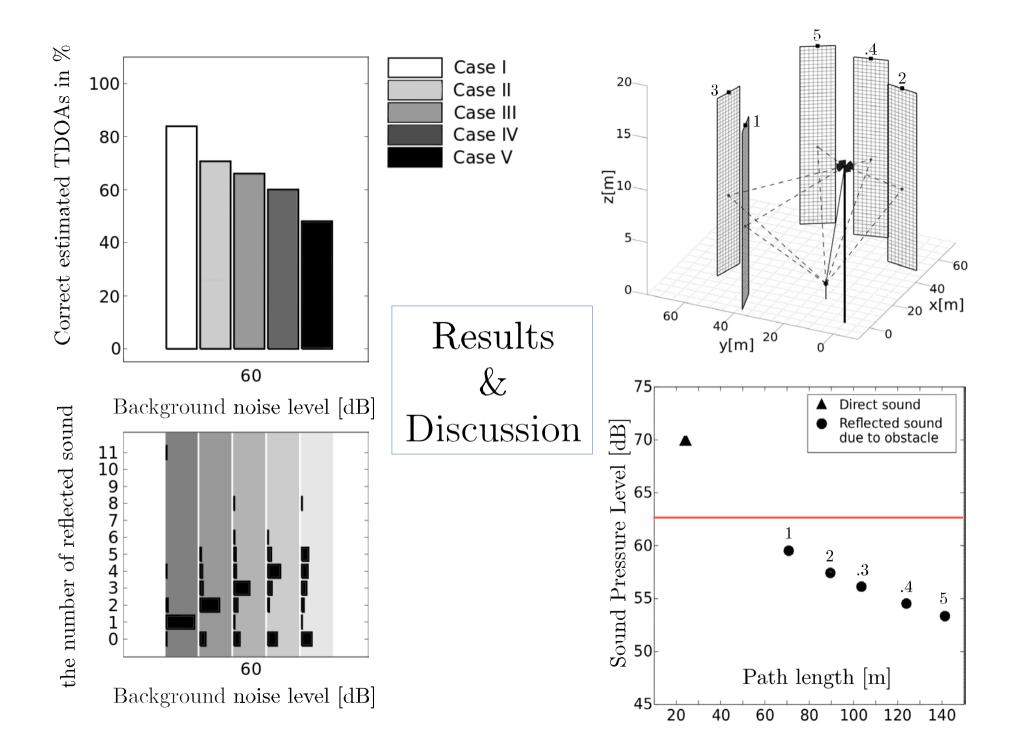


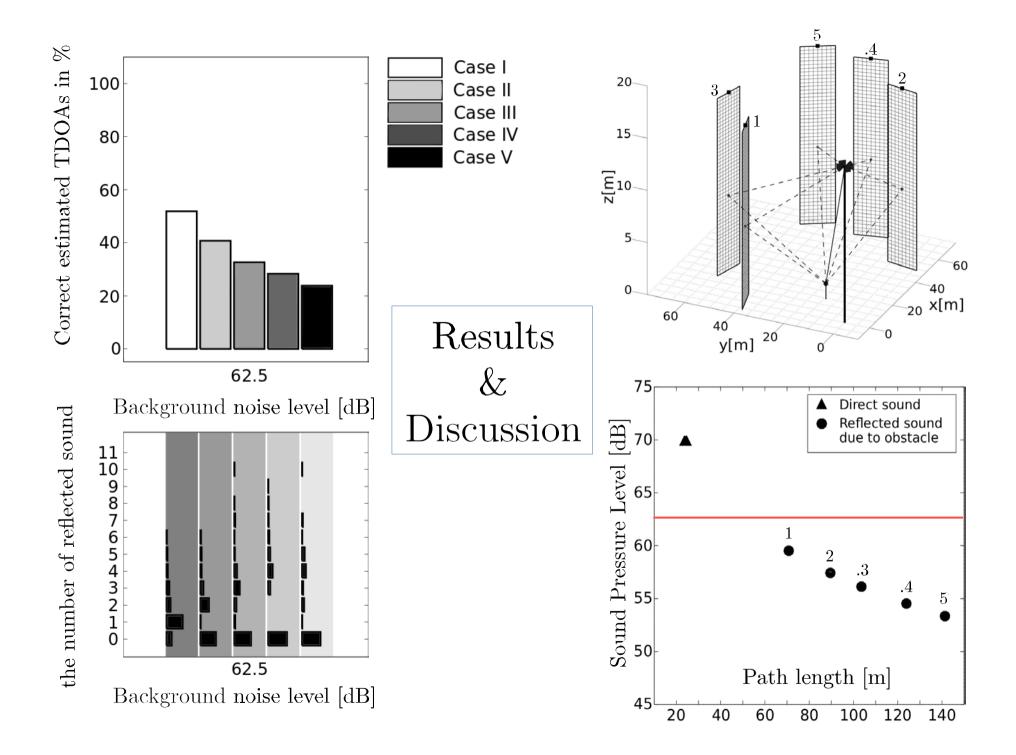




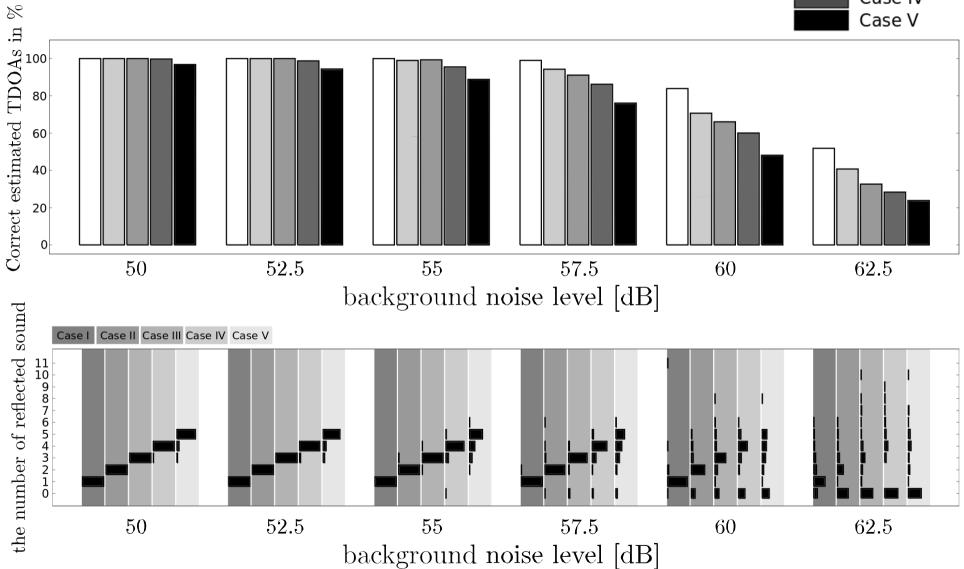












Conclusions

- Intelligent public address system with a network connection is introduced.
- Since the algorithm is simple, implementation on a single board computer is easy.
- Estimation of direct and reflected sounds is proposed.

Future works:

- Experiments in a field.
- Mapping system will be developed.