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# Python-Based Real-Time Direction-of-Arrival Estimation Using Deep Learning

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### 4. Parameters

- Trained with simulated room impulse responses of 6 rooms with dimensions (6 × 6 × 2.7), (5 × 4 × 2.7), (10 × 6 × 2.7), (8 × 3 × 2.7), (8 × 5 × 2.7),
- with reverberation-times  $\in \{0.3 \ s, 0.2 \ s, 0.8 \ s, 0.5 \ s, 0.7 \ s, 0.8 \ s\}$  and 7 positions per room.
- STFT-parameters: 16 kHz sampling frequency, 10 ms hop-length,
- An uniform linear array (ULA) with four microphones captures directional, reverberant sound sources.
- The inter-microphone distance is 8 cm.
- The direction-of-arrival  $\theta \in [0, 180]$  is to be obtained from the microphone signals (one or more sound sources).

## 2. DOA Estimation [1,2]



Compute the STFT-phases of the microphone signals.

32 ms window length, Hann-window.

#### 5. Demonstration



#### Visualization 1

AUDO

ABS

- DNN output of all 37
   DOA classes over
   time (update-rate
   10 ms).
- Output values below0.5 are set to zero.



#### Visualization 2

Polar plot of the 37 DOA classes (update-rate 10 ms).
DNN output averaged over 5 time-frames (72 ms).

- Feed the phases of a single time-frame into a CNN with subsequent feed-forward layers.
- The phases are mapped to 37 DOA classes, which represent an angular DOA resolution of 5 degrees.
- Multiple sources can be simultaneously localized via classification.





# 6. Ask for ...

- ... offsets of the microphones to show robustness.
- varying inter-microphone distances.
- different audio sources (e.g., clapping, clinking, speech, etc.).

[1] Soumitro Chakrabarty and Emanuël A. P. Habets, "Broadband DOA estimation using convolutional neural networks trained with noise signals," in *Proc. IEEE Workshop on Applications of Signal Processing to Audio and Acoustics (WASPAA)*, Oct. 2017, pp. 136–140.

Time-Frame (N) Time-Shuffle

- Training data is simulated with white noise sources convolved with simulated room impulse responses.
- To simulate concurrent sound sources, a time-shuffle is applied to the phase maps of two sources.

[2] Soumitro Chakrabarty and Emanuël A. P. Habets, "Multi-speaker DOA estimation using deep convolutional networks trained with noise signals," *IEEE J. of Sel. Topics in Signal Processing*, vol. 13, pp. 8–21, Feb. 2019.



