SOUND ANALYSIS, SYNTHESIS AND PROCESSING

DAAP Homework

Report on Source Localization Using Delay-and-Sum Beamforming

Objective

The goal of this assignment was to perform acoustic source localization using a uniform linear array (ULA) with 16 microphones. The localization method utilizes the delay-and-sum (DAS) beamforming approach, processing the recorded multichannel data in both time and frequency domains.

Implementation Overview

The code implements the following steps to achieve source localization:

1. System Setup:

- The ULA consists of M=16 microphones spaced uniformly along a length of L=0.45 m.
- \circ The inter-microphone spacing d, is computed as d=L/(M-1).
- $\circ~$ The speed of sound is taken as c=343 m/s, and the sampling frequency is Fs=8000 Hz

2. Signal Processing:

- A custom Short-Time Fourier Transform (STFT) function was implemented to decompose the time-domain signals into time-frequency representations. The STFT is computed using:
 - Window length of 256.
 - Hop size of 128.
 - FFT size of 512.
- Each frame of the signal is multiplied with a Hann window before applying the FFT.

3. Beamforming:

- \circ The DAS beamformer computes the pseudospectrum for each time frame and frequency bin using steering vectors designed for angles θ, ranging from -90° to 90° .
- The steering vectors incorporate the array geometry and the wave properties.

4. Pseudospectrum Visualization:

- The frequency-averaged pseudospectrum is computed and plotted as a 2D map with time on the y-axis and angles on the x-axis.
- The resulting pseudospectrum is normalized and converted to a logarithmic scale for better visualization.

5. **DOA Estimation**:

 The code identifies the Direction of Arrival (DOA) at each time frame by finding the angle corresponding to the maximum power in the pseudospectrum.

6. Visualization:

- o A static plot shows the ULA setup and the time-varying DOAs as arrows.
- A video illustrates the DOA estimation dynamically over time, with arrows pointing to the estimated source positions.

Key Functions

- STFT Implementation (stft_custom): A custom implementation avoids reliance on built-in MATLAB functions. It slices the signal into overlapping frames, applies a window function, and computes the FFT.
- Steering Vector Calculation (steering_vector): This function models the propagation delay across microphones for plane waves arriving from various angles.
- DAS Beamforming (das_beamformer): Computes the beamforming output by projecting the microphone signals onto the steering vector.

Results

1. Frequency-Averaged Pseudospectrum:

- The generated pseudospectrum plot successfully shows the power distribution over time and DOA.
- The logarithmic normalization highlights significant sources effectively.

2. ULA Setup and DOAs:

 The ULA is visualized with microphone positions and DOAs depicted as arrows, indicating source directions dynamically over time.

3. Localization Video:

 The video demonstrates time-varying source localization, showcasing the system's capability to track moving sources.

Challenges and Considerations

- **Wideband Sources**: The DAS beamformer is a narrowband technique. To handle wideband sources, the information was averaged across frequencies.
- **Resolution**: The array length and inter-element spacing limit the angular resolution. For better resolution, larger arrays or smaller microphone spacing are necessary.

Conclusion

The implemented solution adheres to the assignment requirements, providing a robust and modular framework for source localization using ULA. The code generates accurate visualizations and videos, demonstrating the effectiveness of DAS beamforming for real-time acoustic source tracking.