

ACOUSTIC SOURCE LOCALIZATION

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OUTLINE

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- Applications
- Techniques used for Acoustic Source Localization
- Literature survey
- Objectives and Scope
- Beamforming
- Delay and sum Algorithm
- References

INTRODUCTION

- Acoustic source localization is the process of determining the location or direction of origin of a sound in a given environment
- Implemented using an array of sensors or microphones
- The goal is to estimate the position or direction of source origin

Sensor Characteristics

- Microphone Array Configuration
- Number of Microphones
- Microphone Sensitivity
- Frequency Response
- Directional Characteristics
- Noise Characteristics
- Dynamic Range





1. Smart Home and IOT Applications

- By accurately localizing sound sources within the home environment
- These systems can provide personalized experiences, enhance user interaction, and improve the efficiency of home automation tasks

APPLICATIONS

2. <u>Industrial Inspection and Maintenance</u>

- > For **monitoring** and **maintenance** of machinery and equipment
- Helps in detecting abnormal sounds, vibrations, or leaks in industrial processes
- > Can identify potential malfunctions or defects early preventing equipment failures





3. Military and Defense

- Locate enemy threats, such as gunfire, artillery, or unmanned aerial vehicles (UAVs)
- Battlefield surveillance
- Target detection





4. <u>Surveillance and Security</u>

- Detect and locate suspicious sounds, such as gunshots, explosions, or unauthorized intrusions
- Respond quickly to potential **threats** and enhance situational awareness in sensitive areas like airports, military installations, and urban environments





5. <u>Search and Rescue Operation</u>

- Locate individuals in distress, such as lost hikers, earthquake survivors, or victims of natural disasters
- > By **detecting** and **pinpointing the source** of distress signals or calls for help

Techniques used for ASL



1. Conventional Beamforming

ADVANTAGES

- Real time processing
- Robust
- Interpretability

DISADVANTAGES

- Require Prior Information
- Limited in Complex Environment
- Sensitive to Array Geometry

^{1.} K. Rashida, S. Kamal, C. Satheesh Chandran, A. A. Balakrishnan and M. H. Supriya, "Prototype Implementation of Spatial Filtering using Sensor Array," 2019 International Symposium on Ocean Technology (SYMPOL)

Techniques used for ASL



2. Al and ML Techniques

ADVANTAGES

- Adaptability
- Feature Learning
- Flexibility

DISADVANTAGES

- Complexity
- Data Dependency
- Black Box Nature

3. Chung, M.-A.; Chou, H.-C.; Lin, C.-W. Sound Localization Based on Acoustic Source Using Multiple Microphone Array in an Indoor Environment. *Electronics* 2022, *11*, 890.



REFERENCE	METHODOLOGY	RESULTS	FUTURE WORK
Chung, MA.; Chou, HC.; Lin, CW. Sound Localization Based on Acoustic Source Using Multiple Microphone Array in an Indoor Environment. Electronics 2022, 11, 890.	The TDOA is also designed to deal with the problem of delay in the reception of sound signals from two microphone arrays by using the generalized cross-correlation algorithm to calculate the TDOA.	The experimental results prove that the proposed method can detect the sound source and obtain good performance with a position error of about 2.0~2.3 cm and angle error of about 0.74 degrees	Localize moving sound sources under this framework, i.e., how to localize them accurately considering the Doppler effect.



REFERENCE	METHODOLOGY	RESULT	FUTURE WORK
Jekateryńczuk, G.; Piotrowski, Z. A Survey of Sound Source Localization and Detection Methods and Their Applications. Sensors 2024, 24, 68.	The primary objective of this work was to comprehensively delve into the techniques of sound source detection and acoustic localization elucidating their diverse applications across both military and civil domains.	The primary objective of this research is to serve as a valuable resource for selecting the most suitable approach within this domain.	Refinement of Algorithms: refining existing algorithms to improve their precision and reliability. This may involve exploring new signal processing techniques, optimization methods, or machine learning approaches to enhance the accuracy of sound source detection and localization.



REFERENCE	METHODOLOGY	RESULTS	FUTURE WORK
Hassan, Farrukh & Mahmood, Ahmad Kamil & Yahya, Norashikin & khan, Zia & Rimsan, Mohamed. (2021). State-of-the-Art Review on the Acoustic Emission Source Localization Techniques. IEEE Access. 9. 10.1109/ACCESS.2021.309 6930.	Numerous localization techniques, such as Modal Acoustic Emission, Neural Networks, Beamforming, and Triangulation methods with or without prior knowledge of wave velocity, have been presented	A deep learning circular sensor cluster-based solution has the potential to provide a low-cost reliable localization solution for acoustic emission sources	Acoustic Emission signals include noise. Sufficient expertise in signal processing is required to analyze the excellent quality of recorded AE signals for damage localization AE signals may wrongly be interpreted as original signals, resulting in incorrect localization results.



REFERENCE	METHODOLOGY	RESULTS	FUTURE WORK
Kan, Yue et al. "Passive Acoustic Source Localization at a Low Sampling Rate Based on a Five-Element Cross Microphone Array." Sensors (Basel, Switzerland) 15 (2015): 13326 - 13347.	Accurate acoustic source localization at a low sampling rate (less than 10 kHz)	Absolute error of the localization results based on the US-GCC method with the interpolation factor 15 is approximately from 1/15 to 1/12-times that based on the GCC method with the same sampling rate.	Possibility is given for applying the US-GCC method with an interpolation factor of 15 to a small portable system, especially a multitasking micro-embedded system



REFERENCE	METHODOLOGY	RESULTS	FUTURE WORK
Rascon, Caleb, and Ivan Vladimir Meza Ruiz. "Localization of Sound Sources in Robotics: A Review." Robotics and Autonomous Systems 96.C (2017): 184–210. Web.	The main objective of this review is to thoroughly map the current state of the Sound source localization (SSL) field for the reader and provide a starting point to SSL in robotics.	Robot audition modules, such as source separation, and it enriches human-robot interaction by complementing the robot's perceptual capabilities.	There are types of sources other than human speech that can be localized , such as dog barks, security alarms, whistles, body movements, stride pacing, etc. These are all dimensions of a fertile landscape in which the SSL field can grow .



Objectives

• To perform **Acoustic Source Localization** using **Signal Processing** techniques in indoor environment



Scope

- Implement of **Delay and Sum Algorithm** for Acoustic Source Localization
- Implement better Algorithms for Acoustic Source Localization
- Improve Acoustic Source Localization using AI/ML techniques



Beamforming

- Use of an array of antennas or transducers to perform signal processing based on the spatial characteristics of a signal.
- Determine the location of a source signal in space

- Characterize the signals from sources, to distinguish different targets
- Use the signals captured by the microphone arrays to automatically obtain the position and details of the acoustic sources.

^{1.} K. Rashida, S. Kamal, C. Satheesh Chandran, A. A. Balakrishnan and M. H. Supriya, "Prototype Implementation of Spatial Filtering using Sensor Array," *2019 International Symposium on Ocean Technology (SYMPOL)*



Beamforming Algorithms

1. Delay and Sum Algorithm

- Signals are delayed and summed coming from different microphones
- By adjusting delays we can focus on different directions to enhance signals arriving from specific directions



Beamforming Algorithms

2. <u>Minimum Variance Distortionless Response(MVDR)</u>

- **Purpose:** Focus on a desired sound source while minimizing **background** noise.
- Method: Calculates the direction that minimizes overall noise while keeping the desired signal clear.
- Adaptability: Adjusts continuously to changes in the environment to maintain clarity.



Beamforming Algorithms

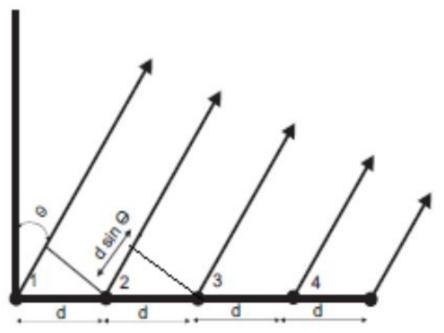
3. Multiple Signal Classification(MUSIC)

- Objective: Identify and locate multiple sound sources in a noisy environment.
- Process: Separates sounds picked up by microphones and analyzes unique arrival patterns.
- Identification: Matches patterns to determine the direction and location of each sound source



- It consists of an array of sensors in a particular configuration
- The outputs from all the sensors are properly aligned and added up to estimate the source direction
- Array configuration should be decided before implementation





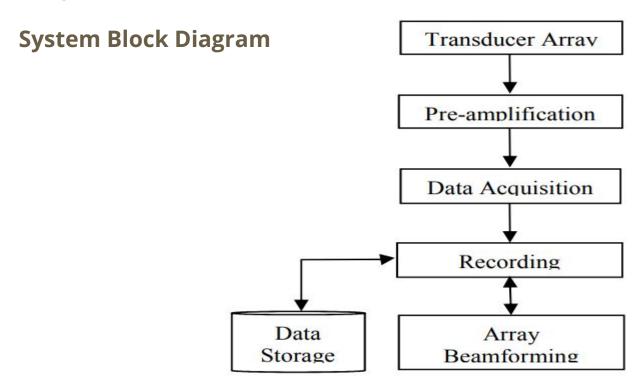
Uniform linear array configuration



$$\tau = \frac{(N-1) \, d \sin \theta}{c}$$

- N is the number of microphones
- **c** denotes the speed of sound in air
- The angle of arrival θ (in degrees)
- d (in meters) between successive elements.





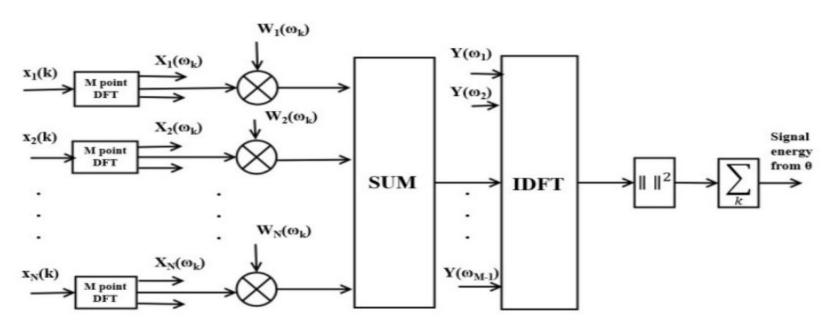


System Block Diagram Explanation

- The transducer unit picks up the signal emanations from both the actual targets as well as the interfering sources
- The pre-processing unit consists of microphone preamplifier
- Data acquisition unit performs analog to digital conversion
- Digital data is recorded on to a storage device
- Array beamforming is performed as a post processing operation, in order to get the spatial information



Frequency domain implementation of wideband beamforming





Let s(t) be the impinging plane wave signals that are received at the origin
of the array coordinate system. Then the signal received at the mth
sensor will be,

$$x_m(t) = s(t - \tau_m)$$

the output of each sensor is sampled every Ts seconds where Ts = 1/Fs,
 where Fs is the sampling frequency



- M point DFT is taken for each sensor output.
- The mth frequency bin consists of an N×1 complex vector Xm
- where m = 0 to M 1

$$X_m \stackrel{\Delta}{=} [X_0 X_1 \dots X_{N-1}]$$

• Wm is the weighting vector corresponding to delay of signal in each sensor

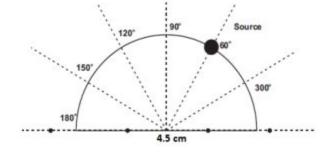
$$Y_m = W_m^H X_m$$



- Now the sum of all N values of each mth frequency bin is evaluated to obtain M frequency samples.
- Finally, inverse DFT is computed to obtain M output samples in the time domain

The output signal energy is calculated for each steering direction from 0

to 180 degree.



^{1.} K. Rashida, S. Kamal, C. Satheesh Chandran, A. A. Balakrishnan and M. H. Supriya, "Prototype Implementation of Spatial Filtering using Sensor Array," 2019 International Symposium on Ocean Technology (SYMPOL)



Inference

- ASL can be done using signal processing using Beamforming techniques and also using ML
- In Beamforming different algorithms are being used which will be chosen according to application
- This approach is computationally efficient and can provide accurate localization results, particularly in scenarios with relatively simple acoustic environments and a small number of sources.
- It requires precise knowledge of the microphone array geometry and the speed of sound in the environment
- Some other techniques, such as adaptive beamforming or machine learning-based approaches, to improve performance in more challenging conditions.

References

[1] K. Rashida, S. Kamal, C. Satheesh Chandran, A. A. Balakrishnan and M. H. Supriya, "Prototype Implementation of Spatial Filtering using Sensor Array," 2019 International Symposium on Ocean Technology (SYMPOL), Ernakulam, India, 2019, pp. 39-46, doi: 10.1109/SYMPOL48207.2019.9005284.

[2] Desai, Dhwani and Mehendale, Ninad, A Review on Sound Source Localization Systems (July 22, 2021)

[3] Chung, M.-A.; Chou, H.-C.; Lin, C.-W. Sound Localization Based on Acoustic Source Using Multiple Microphone Array in an Indoor Environment. *Electronics* **2022**, *11*, 890.

[4] Jekateryńczuk, G.; Piotrowski, Z. A Survey of Sound Source Localization and Detection Methods and Their Applications. *Sensors* **2024**, *24*, 68.

[5] Hassan, Farrukh & Mahmood, Ahmad Kamil & Yahya, Norashikin & khan, Zia & Rimsan, Mohamed. (2021). State-of-the-Art Review on the Acoustic Emission Source Localization Techniques. IEEE Access. 9. 10.1109/ACCESS.2021.3096930.

[6] Kan, Yue et al. "Passive Acoustic Source Localization at a Low Sampling Rate Based on a Five-Element Cross Microphone Array." Sensors (Basel, Switzerland) 15 (2015): 13326 - 13347.

[7] Rascon, Caleb, and Ivan Vladimir Meza Ruiz. "Localization of Sound Sources in Robotics: A Review." Robotics and Autonomous Systems 96.C (2017): 184–210. Web.



THANK YOU