

3D Localization of Sound Source Using Microphone Array

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ABSTRACT: Speaker identification is found to be an important feature of modern-day contactless communication establishment. This presents a unique and urgent need to create an accurate and efficient localizer. This work aims to demonstrate a 3D localization procedure with the help of a GPS/GSM module enabled with a set of mics. A mathematical analysis is performed to find elevation and azimuth angle for the sound source. The integrated GPS system computes the latitude and longitude of the mics and GPS system receive the coordinates of the mic and transmits this value with azimuth and elevation angle to the smartphone interface.

KEYWORDS: Interaural level difference (ILD), Interaural time difference (ITD), Head related transfer function (HRTF), Generalized cross-correlation (GCC), and Time differences of arrival (TDoA).

1. INTRODUCTION

Sonar [1], radar [2], teleconferencing [3], mobile phone location [4], navigation and global positioning [5], earthquake location [6], microphone arrays [7], robots [8], micro seismic events in mines [9], sensor networks [10, 11], human-computer interface [12], speaker monitoring [13], surveillance [14], and sound source tracking [15] are variety of applications, where sound source tracking [15] is used for different applications.

The ongoing pandemic situation demands contactless connection for different essential services such as banking. The voice signal can be used as a very simple base for source localization in contactless identification. Studying energy, temporal, and/or directional features from incoming sound at different microphones and using a suitable model that relates those features with the spatial location of the source of interest can be used to localize acoustic sources.

The following is the flow of this paper: section 2 discusses the relevant literature, section 3 describes the proposed TDE-ILD-HRTF method of source localization, section 4 summarizes our experimental results, section 5 discusses the integration with the Geo-location module, and section 6 concludes the paper.

2. REVIEW OF LITERATURE

Estimating the arrival time differences (TDoA) for the acquired signal and mapping the signals to the source position is the most common localization method. Since the distance between the source and the array is more than ten times the aperture's area, the far-field hypothesis is confirmed. Only if the cross-correlation of all the acquired signals is known then only TDoA can be calculated for two or more than two microphones. The maximum cross-correlations of the signals acquired include the lag that leads directly to the TDoA estimate. The cross-correlation approach provides greater robustness by applying the weighting function in the cross-spectrum. The generalized cross-correlation (GCC) mechanism is this robust technique. Changing the weighting function will result in a different algorithm. A GCC weighting function based on phase transforms (PHAT) has been introduced. Sound source localization using the steered response power (SRP) method is an extension of the GCC technique, and this method will be dominated by classical SRP (C-SRP).

Arrival route (DoA) [16], TDoA or interaural time difference (ITD)-[17], [18], [19], and [20], methods based on ILD-[21], [22], [23], and [24], and Head Dependent Transfer Function are the most popular passive sound source localization methods (HRTF). DoA-based beamforming and subspace approaches require a large number of microphones to approximate narrowband source locations in far-field and wideband source locations in near-field instances.

A variety of localization methods have been proposed for multiple signal estimation, covariance approximation, and signal parameter estimation using rotational invariance techniques. The ILD-based approach, on the other hand, is commonly applicable to [high signal-to-noise ratio (SNR)] [21], [22], [23], and [24] single dominant sound sources in the case of a single dominant sound source. For ILD- or TDE-based methods, three microphones are required for 2D positioning and four for 3D [17], [25, 26], [27], [28], [29]. The arrival angle can only be calculated using HRTF-based methods [30].

However, GCC with an ML estimator is the most widely used tool for TDE. Later, a slew of methods for improving GCC in the presence of noise were proposed. In the last decade, several papers have attempted to use HRTF in conjunction with TDE to estimate elevation and azimuth angle of arrival [31, 32]. In this work we try to explore TDE-ILD-HRTF model for source localization.

3. Proposed TDE-ILD-HRTF Method

3.1 Source Counting Method

One sound signal (source signal) is considered as:

$$s(t) = s(t + T) \quad (1)$$

and two $s(t)$ and $s^l(t)$ signal in high SNR localization area. Therefore equation (1) can be rewrite as:

$$s_1(t) = s^1(t - T_1') + s(t - T_1) \quad (2)$$

$$s_2(t) = s^1(t - T_2') + s(t - T_2) \quad (3)$$

And correlation function is given as

$$R_{s_1 s_2}(\tau) = \int_{-\infty}^{\infty} (s^1(t - T_1') + s(t - T_1)) (s^1(t - T_2' + \tau) + s(t - T_2 + \tau)) dt \quad (4)$$

$$R_{s_1 s_2}(\tau) = R_{1s_1 s_2}(\tau) + R_{2s_1 s_2}(\tau) + R_{3s_1 s_2}(\tau) + R_{4s_1 s_2}(\tau) \quad (5)$$

Using above equations, maximum value for $R_{1v_1 v_2}(\tau)$ is given by $\tau_1 = T_2 - T_1$ and similar expressions follow for $R_{2v_1 v_2}(\tau)$, $R_{3v_1 v_2}(\tau)$ and $R_{4v_1 v_2}(\tau)$. Therefore, four peak values are there in cross-correlation vector and by counting the values of dominant cross-correlation vector, number of dominant and active sources in localization area can be easily found.

3.2 Extension of Dimensions to Three

Half-cylinder reflector is only applicable in 2D cases in the proposed localization system. Instead of half-cylinder, a half-sphere reflector is more effective for 3D sound source localization. Half-sphere reflector allows localization structure to localize sources of sound in 3D case as shown in Figure 1.

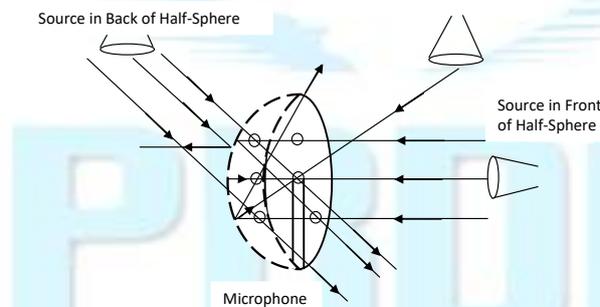


Figure 1 Half sphere as a coarse approximation

4. EXPERIMENTAL OBSERVATIONS

4.1 Experimental Setup

The proposed hardware is implemented using a 'MB800' motherboard and DELTA IOIO LT sound card from M-Audio, with up to 16-bit resolution and 96-kHz sampling frequency, and a B-29L preamplifier with three 'C-417' style microphones, as shown in Figure 2.



Figure 2 Experimental mic setup

A 2-cm-diameter custom reflector is mounted behind the third microphone, as shown in Figure 3.



Figure 3 The proposed 3D sound source localization setup

5. INTEGRATION WITH GEO-LOCATIONING

A set of mics integrated with GPS/GSM module is placed in a small geographical area. The received signal strength of all the mics is monitored using simulation tool and two of these mics are identified which are providing the highest value of sound magnitude.

Figure 4 shows the integration of acoustic-based directionality with GPS device.



Figure 4 Integration of acoustic-based directionality with GPS devices

Table 1 and Table 2 indicate the angle of arrival measurements.

Table 1 Hardware implementations based measured values of the azimuth angle (θ) of Arrival

Real angle ϕ in ($^{\circ}$)	$\phi_{measured}$ in ($^{\circ}$)	Error in ($^{\circ}$)
0	0.18	0.18
15	15.13	0.13
30	29.85	0.15
45	45.18	0.18
60	60.14	0.14
75	74.91	0.09
90	89.83	0.17
105	104.82	0.18
120	120.14	0.14
135	135.11	0.11
150	150.14	0.14
165	165.09	0.09
180	179.93	0.07

Table 2 Hardware implementations based measured values of the elevation angle (θ) of arrival

Real angle θ in ($^{\circ}$)	$\theta_{measured}$ in ($^{\circ}$)	Error in ($^{\circ}$)
-10	-10.18	0.18
-5	-5.16	0.16
0	0.05	0.05
15	14.15	0.15
30	29.91	0.09
45	45.19	0.19
60	60.13	0.13
75	75.11	0.11
90	89.82	0.18

The integrated GPS system computes the latitude and longitude of the mics and GPS system receives the coordinates of the mic and transmits this value with azimuth and elevation angle to the smart phone interface. Figure 5 shows/locates the user in larger geography.



Figure 5 Integration with Google maps to increase usability

6. CONCLUSION

In this paper, the designed project aims to localize the 3D sound source (direction only) without recognizing the speaker, i.e., who is speaking and the content of the speech. To demonstrate this 3D localization procedure, a set of mics integrated with GPS/GSM module are placed in a small geographical area. The received signal strength of all the mics is monitored using simulation tool and two of these mics are identified which are providing the highest value of sound magnitude. A mathematical analysis is done after this to calculate the approximate both elevation and azimuth angle for the sound source with respect to the two identical mics (with highest output). The

integrated GPS system computes the latitude and longitude of the mics and GPS system receives the coordinates of the mic and transmits this value with azimuth and elevation angle to the smart phone interface.

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