Subband Weighting for Binaural Speech Source Localization

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Abstract

We consider the task of speech source localization from a binaural recording using interaural time difference (ITD). A typical approach is to process binaural speech using gammatone filters and calculate frame-level ITD in each subband. The ITDs in each gammatone subband are statistically modeled using Gaussian mixture models (GMMs) for every direction during training. Given a binaural test-speech, the source is localized using maximum likelihood (ML) criterion. In this work, we propose a subband weighting scheme where subband likelihoods are weighted based on their reliability. We measure the reliability of a subband using the average frame level localization error obtained for the respective subbands. These reliability values are used as the weights for each subband likelihood prior to combining the likelihoods for ML estimation. We also introduce non-linear warping of these weights to accommodate and analyze a larger space of possible subband weights. Experiments on Subject_003 from the CIPIC database reveal that weighting the subbands is better than the unweighted scheme of combining likelihoods.

Index Terms: gammatone filters, interaural time difference, warping function

1. Introduction

Machine localization of speech sources is essential for a wide range of applications, including human-robot interaction, surveillance and hearing aids. Robot sound localization algorithms have been proposed using microphone arrays with varied number of microphones [1–6]. However, humans have an incredible ability to localize sounds with just two ears. The major cues that help in localization are interaural time difference (ITD) and interaural level difference (ILD). These cues can be captured by the head-related transfer function (HRTF) [7]. Its time domain equivalent is the head-related impulse response (HRIR). Algorithms inspired by binaural localization of humans would extract these features from the input signals [8–22]. To account for the time and frequency variability of these cues, time-frequency (TF) representations of the binaural signals are used. One of the most common time-frequency representations is the Short-Time Fourier Transform (STFT) [10,12,15,17,19]. Another approach is to use gammatone filters [23] where, unlike STFT, the subband width and spacing are not uniform [8,11,13,14,16,22,24]. The use of gammatone filters is inspired by the filter structure of the cochlea in human ears. In this work, we use gammatone filters to preprocess the binaural signals.

May et al. [13, 14] and Woodruff et al. [16] use Gaussian mixture models (GMMs) to model ITD, ILD and their joint distribution (ITLD) for each gammatone subband in each direction. Then, for a test-speech, log-likelihood integration is performed across the TF plane and the direction with the maximum likelihood is picked as the direction of arrival (DoA). Ma et al. [22] use DNNs to map the cross-correlation function, instead of ITDs, of each TF bin to the posterior distribution (PD) of DoAs. PDs across the TF plane are averaged similar to the likelihood integration in GMM based methods and the direction with the maximum posterior probability is chosen as the DoA. In this work, we consider only ITD and extend the GMM based method of May et al. [13] by investigating the localization accuracy of each subband. DoA estimation can be treated as a classification problem where, given a feature (ITD), the latent class (DoA) needs to be inferred. For a subband, each direction will have its own distribution of ITD. Higher discrimination among distributions of different directions results in a better classification accuracy. Hence, subbands with a high level of discrimination are more reliable than the ones with a low level of discrimination. Addition of noise could decrease this discrimination and lead to a decrease in localization accuracy. We hypothesize that weighting the subband likelihoods based on their reliability can improve localization accuracy. Using localization error as a measure of discrimination, we find a measure of reliability for each of these subbands. These reliability values are then used as the subband weights. To account for a larger space of weights and possible changes in the reliability of subbands with SNR, we propose to warp the weights using a non-linear warping function. From the localization errors obtained for all possible weights at each SNR, we select one set of weights that performs best on all SNRs combined. Experiments with Subject_003 from the CIPIC database [25] reveal that the weighted subband localization scheme works better than the unweighted scheme. It also turns out that the best weights obtained for each SNR performs marginally better than the SNR independent weights under the simulated additive white gaussian noise (AWGN) conditions.

2. Binaural Cue Extraction and Localization

DoA estimation consists of the steps shown in figure 1. We provide the details of these steps in the following subsections.

2.1. Gammatone Filters

The binaural signals are processed through N=32 fourth order gammatone filters. Their center frequencies are equally distributed with respect to the equivalent rectangular bandwidth (ERB) scale between 80Hz and 5kHz, starting with 80Hz and ending with 4.6kHz. This range primarily covers the entire speech spectrum. To approximate the neural transduction process of the inner hair cells, the outputs of the gammatone filters are halfwave rectified and square-root compressed [13]. The resulting outputs of the left and right channels of the ith subband are denoted by li and ri.
2.2. ITD Extraction

Frame-level ITD in each subband is then calculated using normalized cross correlation (NCC) \[8, 13\] between \(i\) and \(j\) with a rectangular window of length \(W\) and shift of length \(W/2\). \(\tau_{i,j}\) is the ITD of the \(i\)th subband in the \(j\)th frame and is given by

\[
\tau_{i,j} = \arg\max_{\tau} C_{i,j}(\tau),
\]

where \(C_{i,j}\) is the NCC function. In addition to this, exponential interpolation is used to obtain fractional delays \[13\].

2.3. GMM Parameter Estimation

GMMs are trained on clean speech ITDs of each subband in each direction. As shown in \[13\], ITD distributions of high-frequency subbands have multiple modes. Hence, the need for GMMs. Information theoretic criteria such as Akaike Information Criterion (AIC) [26] and Bayesian Information Criterion (BIC) [27] are used to evaluate the optimum number of components in each GMM. The optimal number of components is obtained using AIC as well as BIC. The lower number between the two is chosen as the optimal number of components.

2.4. Likelihood and Localization

Suppose we consider \(D\) directions. Then \(\lambda_{i,d}\) is the set of GMM parameters for the \(i\)th subband in the \(d\)th direction where \(d\) ranges from 1 to \(D\). \(\tau_{i,j}\) denotes the ITD of the \(i\)th subband in the \(j\)th frame. Then, \(p(\tau_{i,j}|\lambda_{i,d})\) is calculated for \(i, d\). Given a test binaural speech, May et al. \[13\] combined the likelihoods of all the subbands, for each \(d\) to obtain a single likelihood for each direction. And then, the direction with the maximum likelihood (ML) is chosen as the DoA estimate in the \(j\)th frame.

\[
\text{DoA}_j = \arg\max_{d \in \{1, \ldots, D\}} \sum_{i=1}^{N} \log p(\tau_{i,j}|\lambda_{i,d})
\]

The DoAs from multiple number of frames (\(n\)f) are pooled to obtain the DoA with the maximum frequency of occurrence.

3. Proposed Subband Weighting

We propose a weighted maximum likelihood (WML) method in which the DoA estimate in the \(j\)th frame is given by

\[
\text{DoA}_j = \arg\max_{\lambda_{i,d}} \sum_{i=1}^{N} w_i \log p(\tau_{i,j}|\lambda_{i,d}),
\]

where \(w_i\) is the weight corresponding to the \(i\)th subband. With the motivation to weight subband likelihoods based on their reliabilities, the method to obtain these weights is described in the following subsections.

3.1. Subband Reliability

Reliability of a subband is measured as a quantity that is inversely proportional to the average localization error of that subband. To calculate the localization error of each subband we need the frame level direction estimates of each subband.

\[
\text{DoA}_{i,j} = \arg\max_{\lambda_{i,d}} \log p(\tau_{i,j}|\lambda_{i,d}),
\]

where \(\text{DoA}_{i,j}\) is the DoA estimate of the \(i\)th subband in the \(j\)th frame. The frame level estimates are then pooled over \(n\)f frames to obtain the final DoA. Figure 2(a) shows the ITD distributions of 25 azimuthal directions in 4 different subbands. It can be seen that subband at 80 Hz has a lot of overlap among the distributions unlike 518.8 Hz whose distributions are well separated. This means that subband at 518.8 Hz is more reliable than 80 Hz. In a previous work \[24\], we evaluated the subband localization errors for clean speech using speech segments of duration \(n\)f = 100. However, here we use \(n\)f = 1. This is equivalent to sampling ITDs from each of the trained GMMs and finding the localization error due to the samples that are incorrectly localized. Hence, choosing \(n\)f = 1 is a better way of measuring discrimination amongst GMMs in a subband. The average localization errors of subbands obtained using \(n\)f = 1 is shown in figure 2(b) (top row). As hypothesized, subband at 518.8 Hz has a localization error lower than that of 80 Hz.
3.2. Non-Linear Warping

In the previous section, reliabilities are obtained using the subband localization errors of clean speech. The relative reliabilities of different subbands might vary in the presence of noise and it can also be SNR dependent. Also, $R_i = T - S_i$ is one way, amongst many other ways, of obtaining reliability. So, to account for these possibilities, we introduce non-linear warping of the obtained reliabilities so as to span over a wide range of possible reliabilities which are used to compute the subband weights. The weight computation and warping function used are defined as follows,

$$w_{\alpha,T}(i) = \frac{2T}{\pi} \arctan \left( \frac{1 - \alpha}{1 + \alpha} \tan \left( \frac{R_i \pi}{2T} \right) \right),$$

where $\alpha$ is the warping parameter and $w_{\alpha,T}(i)$ is the weight of the $i$th subband. Top row of figure 2(b) shows the warping functions for different values of $\alpha$. The corresponding warped weights are shown respectively in the bottom row. It should be noted that choosing $\alpha = -0.9999$ is equivalent to weighting all the subbands equally. However, $\alpha = 0.9999$ almost corresponds to choosing just one subband corresponding to the lowest localization error, i.e., the highest reliability. So, given a pair of parameters $(\alpha, T)$ and the subband errors $(S_i)$, the corresponding weights are calculated as per the following steps.

Step 1: $S_i \leftarrow S_i - \min(S_i)$.

Step 2: $R_i \leftarrow T - S_i$.

Step 3: $w_{\alpha,T}(i) \leftarrow \text{Substitute } R_i, \alpha \text{ and } T \text{ in eqn. (5)}$.

Step 1 is essential to ensure that the subband with the minimum error (maximum reliability) is always given the maximum weight equal to $T$.

3.3. $\alpha$ Sampling

$\alpha$ has a range of $(-1, 1)$. However, uniformly choosing alphas in this range is not reasonable as the rate of change of weights is not linear with $\alpha$. We define the rate of change of weights as

$$v(\alpha,T) = \lim_{\delta \to 0} \frac{1}{2\delta^2} \sum_{i=1}^{N} (w_{\alpha+i\delta,T}(i) - w_{\alpha-i\delta,T}(i))^2,$$

where $v(\alpha,T)$ is the rate of change of weights at $(\alpha,T)$. We then integrate this function with respect to $\alpha$ to get the cumulative change, $f(\alpha,T)$. Figure 3 shows $f(\alpha,T)$ for $T = 14.42$. $f(\alpha,T)$ is then used to select different values of $\alpha$. $f(1,T)$ is equal to the total cumulative change. To obtain a sample of $\alpha$ we find the $\alpha$ such that $f(\alpha,T) = \frac{1}{2}(f(1,T) - f(0,T),...,f(1,T))$. The motivation is to sample more $\alpha$ values in the regions with higher rate of change of weights. In figure 3 it can be seen that up to $\alpha \approx 0.8$ there is hardly any increase in $f(\alpha,T)$. However, it then increases steeply. As can be seen in figure 3, there is a higher density of samples close to $\alpha = 1$.

3.4. Best SNR independent weights

For each SNR, we calculate the average localization error of WML using all possible $(\alpha, T)$ pairs. Let $E_{\alpha,T}(k)$ be the average localization error of WML at the $k^{th}$ SNR obtained using $(\alpha, T)$. Then, the minimum localization error for the $k^{th}$ SNR is given by $E_{\text{min}}(k) = \min_{\alpha,T} E_{\alpha,T}(k)$. Now, to obtain the best SNR independent weights (BSIW), we need to obtain the $(\alpha, T)$ pair which performs best over all SNRs combined. The best $(\alpha, T)$ pair is given by

$$(\alpha_{\text{best}}, T_{\text{best}}) \leftarrow \arg\min_{\alpha,T} \sum_{k=1}^{nSNR} (E_{\alpha,T}(k) - E_{\text{min}}(k))^2,$$

where $nSNR$ is the total number of SNRs considered.

4. Experiments and Results

4.1. Database

Speech from TIMIT database [28] is used for all experiments. Binaural speech is simulated using HRIRs of Subject_003 from the CIPIC database [25].

4.2. Experimental Setup

**Binaural speech data preparation:** Localization experiments are performed only in the frontal horizontal plane. The CIPIC database consists of HRIRs of 25 directions in the frontal horizontal plane. Speech from the TIMIT database has a sampling frequency of 16kHz, whereas CIPIC HRIRs are sampled at 44.1kHz. Therefore, speech is upsampled to 44.1kHz and then filtered through the HRIRs to obtain binaural speech corresponding to each direction.

**ITD extraction & GMM parameter estimation:** ITDs are calculated using eqn. (1) with a frame duration of 20msec ($W = 882$) and a frame shift of 10msec ($W_s = 441$). GMMs are trained using frame-level ITDs that are computed from a training binaural speech of duration 10sec. This provides 1000 frames to train each of the 800 (25 directions × 32 subbands) GMMS. EM algorithm [29] with random initialization is used for parameter estimation. As described in Section 2.3, AIC and BIC are used to compute the optimal number of Gaussian components. However, the maximum number of components is restricted to 20.

**Localization error:** Let $\phi$ be the actual azimuthal angle and $\hat{\phi}$ be the estimated angle. Then the localization error is $\epsilon = |\phi - \hat{\phi}|$. Average localization error is obtained by taking the mean of the localization errors in $nS$ different binaural speech segments. We use $nS = 1000$ for calculating subband reliability. In all other experiments we consider $nS = 180$. The localization errors are calculated in degrees.

**Subband reliabilities:** Subband reliabilities are calculated using average localization error of each subband as per the procedure outlined in section 3.1. Bottom row of figure 2(b) shows the obtained reliabilities. Subband with center frequency 518.8 Hz has the highest reliability as it has the least average localization error equal to 0.028°.

**$\alpha$ Sampling & $T$:** Thresholds are considered from 14.42 to 40 in steps of 0.5. It has been observed that the sampled $\alpha$ values do not change considerably with change in $T$. Hence, a set of 100 $\alpha$ values common to all $T$ has been sampled using $T = 14.42$. A dense set of $\alpha$ starting from -0.9999 to 0.9999 with a step of 0.0001 have been considered to obtain $f(\alpha,T)$. $\delta = 0.00005$ has been used to obtain the rate of change function $v(\alpha,T)$. $1/2\delta$ factor in eqn. (6) has been omitted while calculating the rate of change as it is a common factor and does not affect the sampling process. The 100 sampled $\alpha$ values are shown in figure 4(a). They start with -0.9998 and end with 0.9999.
4.3. Results and Discussion

**Best weights:** We first evaluate the localization accuracy of WML for all pairs of $(\alpha, T)$ at different SNRs. We consider SNRs varying from -20dB to 40dB in steps of 2 by adding AWGN. The SNR is relative to each channel. We also consider different durations of test speech i.e. $nf = 10, 40, 70$ & 100. Localization errors at every pair of $(\alpha, T)$ for $nf = 10$ at SNR = -12 dB is shown in the 3D surface plot in figure 4(b). Figure 4(c) shows the surface profile for a fixed $T = 39.92$. Figure 4(d) zooms into the region with maximum density of $\alpha$ samples. It can be seen that this region has the lowest localization errors.

The best $(\alpha, T)$ pairs for each SNR for each $nf$ are then obtained as shown in figure 5. Now we obtain the best SNR independent $(\alpha, T)$ pair using the procedure outlined in Section 3.3. Similarly, we also find the worst SNR independent $(\alpha, T)$ pair i.e., the $(\alpha, T)$ pairs that maximize the quantity on the right hand side of eqn. (7). The best and the worst SNR independent $(\alpha, T)$ pairs obtained for each $nf$ are shown in figure 5 and in the table below.

<table>
<thead>
<tr>
<th>nf</th>
<th>$\alpha_{best}$</th>
<th>$T_{best}$</th>
<th>$\alpha_{worst}$</th>
<th>$T_{worst}$</th>
</tr>
</thead>
<tbody>
<tr>
<td>10</td>
<td>0.9996</td>
<td>23.42</td>
<td>-0.9998</td>
<td>15.92</td>
</tr>
<tr>
<td>40</td>
<td>0.9997</td>
<td>17.92</td>
<td>-0.9998</td>
<td>15.92</td>
</tr>
<tr>
<td>70</td>
<td>0.9998</td>
<td>26.42</td>
<td>-0.9998</td>
<td>38.42</td>
</tr>
<tr>
<td>100</td>
<td>0.8152</td>
<td>28.92</td>
<td>-0.9998</td>
<td>38.92</td>
</tr>
</tbody>
</table>

The weights corresponding to $(\alpha_{best}, T_{best})$ and $(\alpha_{worst}, T_{worst})$ of $nf=10$ are shown in figure 4(e). As seen in the figure, using the best weights is almost equivalent to selecting just the most reliable subband i.e. subband with center frequency 518.8 Hz. On the other hand, using the worst set of weights uniformly weights all the subbands. This is equivalent to ML.

**Performance Evaluation:**

The following schemes are evaluated: WML with reliabilities in Section 3.1 as weights (WML($\alpha=0$)), WML with the best SNR independent weights (WML-BSIW), WML with the worst SNR independent weights (WML-WSIW) and WML with the best SNR specific weights (WML-BSSW). As seen in figure 6, WML-BSIW performs much better than ML especially at very low SNRs with an average localization accuracy improvement of upto 30° to 40°. The performance of WML-WSIW is close to ML. $\alpha_{worst}$ for all values of $nf$ is equal to -0.9998. Choosing $\alpha = -0.9998$ is almost equivalent to uniformly weighting all subbands which is the same as ML. This shows that uniformly weighting all subbands has the lowest performance among all the weights considered. Another important observation is that WML-BSIW performs as good as WML-BSSW at all SNRs. This suggests that in the presence of AWGN, choosing a fixed set of weights over all SNRs yields a performance comparable to using SNR specific weights.

5. Conclusions

We propose a weighted Maximum Likelihood (WML) method for binaural speech source localization. We present a measure of reliability of each subband obtained by using frame level localization error which measures the discrimination of the trained GMMs in each subband. This reliability is, in turn, used as the weights with the inclusion of non-linear warping to account for changes in reliabilities with SNR and also to span a larger space of possible weights. Experimental results with the best set of weights show that WML performs better than ML. It has also been observed that WML-BSIW performs as good as WML-BSSW suggesting that in the presence of AWGN, selection of SNR specific weights is not necessary. It would be interesting to extend this weighting scheme to other kinds of noisy conditions including diffuse noise and reverberation.
6. References


