Trial-Based Calibration for Speaker Recognition in Unseen Conditions

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Abstract

This work presents Trial-Based Calibration (TBC), a novel, automated calibration technique robust to both unseen and widely varying conditions. Motivated by the approach taken by forensic experts in speaker recognition, TBC delays estimating calibration parameters until trial-time when acoustic and behavioral conditions of both sides of the trial are known. An audio characterization system is used to select a small subset of candidate calibration audio samples that best match the conditions of the enrollment sample and a subset that resembles the test conditions. Calibration parameters learned from the target and impostor trials are generated by pairing up these samples and then used to calibrate the score output from the SID system. Evaluated on a diverse, pooled collection of 11 different databases which 14 distinct conditions, the proposed TBC outperforms traditional calibration methods and obtains calibration performance similar to having an ideally matched calibration set.

1. Introduction

Calibration is an important aspect in the usability of speaker identification (SID) systems. Calibration aims to transform scores to log-likelihood ratios (LLR) so that a single identification score can be meaningfully interpreted. For well-calibrated scores the optimal decision threshold for a certain cost function, given by a linear combination of the probability of miss and probability of false alarm, can be theoretically determined using Bayes decision theory.

In this work, we explore the problem of calibration when the trial conditions are variable. We wish to obtain a set of calibrated scores for which the optimal decision threshold computed for each pair of enrollment and test conditions is independent of these conditions. For example, we want the optimal thresholds to be the same for both telephone channel and microphone channel conditions, or for mixed channel conditions. Even the most accurate SID systems, if left un-calibrated or calibrated without regard for trial conditions, require a variety of condition-specific thresholds if optimal decisions are required within each condition.

Techniques developed to cope with varying conditions, such as Discriminative Probabilistic Linear Discriminant Analysis (DPLDA), incorporate information regarding conditions into the calibration parameters. Information such as predicted gender, language and channel can be extracted via Universal Audio Characterization [1] and represented as a low-dimensional vector of class posteriors. These dynamic calibra-

tion methods attempt to adjust the calibration shift and bias according to conditions observed in the trial. As shown in this work, DPLDA fails to adapt well to a score space not well covered during training.

The objective of this work was to investigate methods of calibrating scores such that a single threshold could be defined for multiple trial conditions to optimize the desired operating point. For this initial study, we focused on calibrating scores to minimize the calibration loss, computed as the difference between the actual cost and the minimum cost, across a set of 14 distinct conditions in a data set supplied by the Federal Bureau of Intelligence (FBI). For this, we focused on an operating point where misses and false alarms are weighted equally.

2. Universal Audio Characterization

Universal Audio Characterization (UAC) is a technique that attempts to represent the conditions of a speech signal in terms of a small dimension vector of class posteriors. Previous work in [1, 2] that accounts for these class posteriors has shown improved calibration performance.

We utilize a Gaussian Backend (GB) to extract side information in which each class of interest is modeled using a single Gaussian. I-vectors are used as input with corresponding output being a vector of likelihoods for each class (in the case of DPLDA calibration, these likelihoods are log-normalized to obtain posteriors assuming equal priors for all classes). We use a GB for each category in which discrimination is useful (for instance, language or channel) after which we concatenate side information vectors for each audio file.

3. Existing Calibration Methods

Calibration aims to transform scores to log-likelihood ratios (LLR) so that a single identification score can be meaningfully interpreted.

Common to all calibration techniques is the need to learn a set of calibration parameters (typically a scale and shift) from a development set. The development set contains both target and impostor scores representative of the conditions expected to be encountered during end use of the system.

Calibration methods considered in this work include simple Logistic Regression [3] and the more complex DPLDA [4] which takes into consideration meta data extracted via UAC. Alternate techniques include Neural Networks and SVMs [5], where their application in the context of heavily-degraded speech has been effective.

3.1. Logistic Regression Calibration

Perhaps the most commonly used calibration technique in the field of SID is Logistic Regression [3]. A calibration model

(shift and bias) is learned using linear logistic regression from a large pool of trials deemed to be representative of the end-use conditions. In many cases, a single model is trained using all development data. This approach optimizes calibration globally for all conditions in the development data. The resulting parameters, though, might not be optimal when performance is computed for each condition separately, specially for conditions not seen during training.To surmount this problem, a separate logistic regression model can be trained for each condition to optimize calibration for each of them. However, this requires prior knowledge of the conditions. Furthermore, conditions not seen during training of the calibration model cannot be handled in this case.

3.2. DPLDA Calibration

Discriminative Probabilistic Linear Discriminant Analysis (DPLDA) [4] takes into account meta- or side-information extracted from each side of the trial to determine the best shift and bias for the trial score. DPLDA attempts achieve robustness to unseen conditions by assuming that such conditions can be made of conditions that were observed during training. That is, a segment spoken in a noisy and reverberant environment for which no training data was available may appear 80% noisy and 20% reverberant to a system that was trained with noisy and reverberant speech as separate classes. Limitations of side-information and the subsequent DPLDA model include unpredictability when salient acoustic and voice conditions not modeled a priori are present in the trials. Although able to cope with multiple trial conditions, the DPLDA model also struggles to accommodate unknown trial conditions (as demonstrated in Section 7).

4. Trial-Based Calibration

In forensic speaker verification, conditions of each side of the trial are first determined and used to construct target and impostor trials from candidate calibration data to closely match the trial conditions. These trials provide probability distributions of target and impostor trials, which are then used to calibrate the score to a log likelihood ratio using Bayes theory.

To address the issue of mismatch between calibration data and evaluation data in an automated manner, we propose an approach motivated by forensic experts in SID. Termed trial-based calibration (TBC), the appraoch is based on standard calibration training technique used in global calibration (logistic regression), however, the decision regarding the selection of data from which to learn calibration parameters is postponed until trial time. At this point, conditions of both sides of the trial can be extracted via UAC and used to select a small subset of highly relevant calibration data to produce trial-specific calibration parameters.

This metadata is then used to rank a set of candidate calibration segments according to similarity to each trial side. Specifically, the low-dimensional UAC vectors for the candidate calibraiton segments are first rank normalized. The UAC vectors from each side of the trial are replaced by the rank for each class against the calibration segments. The Euclidean distance of normalized trial and calibration segments is indicative of condition similarity. The most relevant calibration segments are then taken to provide a defined number of target trials (approximately 500) by sweeping the Euclidean distances. Note that this may results in a different number of enrol segments compared to test segments for the calibration set. Calibration parameters
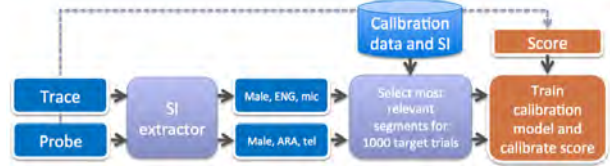

Figure 1: Flow diagram of Trial-based Calibration (TBC).

Table 1: 14 Condition Evaluation Corpus.

Cond.	Chan(s)	Lang(s)	N Spks	Source Corpora
01	Mic	Arabic	240	Pan-Arabic
02	Mic	Arabic	422	Pan-Arabic, LASRS
03	Mic	Cross	179	LASRS
04	Tel	English	225	NIST99
05	Tel	English	467	LASRS, NoTel, NIST99
06	Tel	Cross	597	CrossInt, LASRS
07	Cell	English	62	NoTel
08	Cell	Cross	460	CrossInt
09	Mic, Tel	English	645	CrossInt, LASRS
10	Mic, Tel	Cross	768	CrossInt, LASRS
11	Mic, Cell	English	460	CrossInt, LASRS
12	Mic, Cell	Cross	632	CrossInt
13	Mic, Cell	English	51	NoTel
14	Mic, Cell	Cross	460	CrossInt

are learned through simple logistic regression using this small, highly relevant calibration dataset with which the trial score is transformed (calibrated). Figure 1 depicts the TBC process.

One of the major shortfalls of the DPLDA approach is the emphasized LLRs during calibration for conditions in which very few or no trials were observed during training. The proposed TBC approach is expected to provide additional robustness over DPLDA in these cases due to the selection of the closest set of segments to the trial sides and enforcing a minimum number of target trial scores from which to learn parameters. In the instance that trial conditions are completely absent from the candidate calibration segments, the selection of the TBC calibration set could be seen as an improvement on random (and therefore, over the standard linear regression model) as the most relevant segments are selected via ranking.

5. Data Sets

5.1. Evaluation Data

The evaluation corpus was supplied by the Federal Bureau of Intelligence (FBI) and consists of 14 distinct conditions including same/cross channel and same/cross language trials from 5856 unique segments. Table 1 details these conditions, the corpora from which they were sourced and the number of speakers involved. Note that when reporting results, comparisons for conditions 07 and 13 will have limited statistical significance due to the limited speaker count. The source data in table 1 contains both matched and cross language trials. The second language

Table 2: Characteristics of the Large Variability Dataset.

Lang.	#Seg	#Spkr	Chan(s)	Source Corpora
Arabic	45	6	tel	SRE,RATS
Chinese	48	8	tel	SRE
Dari	74	13	tel	RATS
Pashto	160	25	tel	RATS
Urdu	136	21	tel	RATS
Other	73	20	tel	SRE,RATS
English	902	89	tel, clean / noisy / reverb mic	SRE

of all cross-language trials is English. First language (L1) trials for the LASR corpus (Beck et al 2004), are in Korean, Spanish and Arabic. For the CrossInt corpus, first language trials are in several of the languages of India, depending on what the first language of the participant was, including Hindi, Gujarati, Bengali, Marathi, Tamil, Kannada and Telugu. The NoTel corpus (Godin, et al 2013) contains telephone recordings from naturally noisy locations in Indian accented English. Both the NoTel and CrossInt corpora were collected by Appen for speaker recognition research, The LASR coprus was collected by BAE Systems. This set of corpora were selected for calibration research in order to represent a very wide range of different conditions, collection sources, environments, languages and channels.

5.2. Calibration Data

Matched Data: A small collection of 1503 segments from the NIST and Fisher corpora of speech data was assembled as an initial held-out dataset. Data was chosen with the goal of matching or approximating conditions in the FBI-provided corpus, although certain (cross-language) trial conditions and languages could not be represented. Both telephone and microphone channels were represented with speakers in most languages offering cross-channel trials. Table 2 details the characteristics of this dataset. The segments provided 10736 target trials and 2.1 million impostor trials from which calibration parameters could be learned.

Large Variability Data: This dataset was collected to have a wide range of conditions and sources without regard for the conditions and channels in the FBI dataset. The intention here was to develop a general calibration set suitable for use in a deployed system where data variability will be higher than in research corpora. This will provide a means of measuring robustness to imperfectly matched conditions.

Data was sourced from NIST SRE corpora, clean data from the DARPA RATS SID task [6] (trimmed to 120 seconds of audio), and artificially reverberated and noisy speech data. This data was split into a calibration set of 1468 segments and a training set of the remaining 7k segments. The latter was used to train the SI extractor because disjoint sets were found to work better for all calibration techniques. There were 12930k target trials and 2.1 million impostor trials from the calibration set. Table 2 details the various conditions and languages in these datasets. Languages listed as other include Farsi, Hindi, Russian, Spanish, Thai, and Vietnamese.
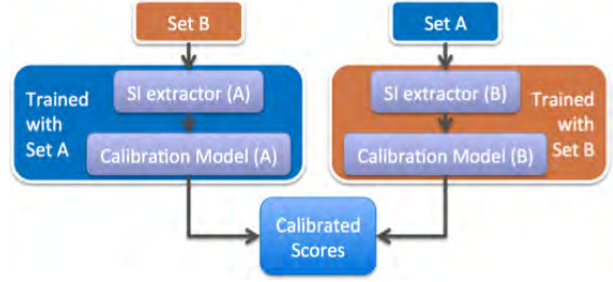

Figure 2: Cross-validation to replicate the ideal matching of evaluation and calibration training data.

Table 3: Comparing the relative improvement of DPLDA over global calibration with different UAC classes when using matched evaluation and calibration data via cross-validation.

UAC Class(es)	Rel. Cllr Improvement
SNR	1%
Channel	13%
Language	40%
Language+Channel	44%
Language+Channel+SNR	32%
Conditio-Specific Calibration	42%

6. Experimental Protocol

The system evaluated in this work was the gender-independent MFCC system consisting of fast, noise-robust voice activity detection [2], a 2048 Gaussian Universal Background Model (UBM), 600 dimensional i-vector subspace and 200D reduction of i-vectors via Linear Discriminant Analysis (LDA) before scoring with a gender-independent probabilistic linear discriminative analysis (PLDA) model [7].

System performance or accuracy is measured in terms of EER. Calibration performance is measured in terms of cost of the likelihood ratio (Cllr), and calibration loss (Closs). Cllr indicates score calibration across all operating points along a detection error tradeoff (DET) curve. In contrast, Closs indicates the systems miscalibration at a particular operating point. For this work, the calibration goal is equal costs for miss and false alarm errors. Closs is calculated as the difference between the minimum cost (assuming perfect calibration) and actual cost. Note that Cllr is a more stringent metric and is not always correlated with Closs. For all metrics, a lower value is more desirable.

7. Results

This section details the evaluation of different calibration approaches and draws conclusions on how effectively they accomplish the goals of this work. Throughout this section, the calibration data varies from closely matched to unmatched with regard to the evaluation data. This variation serves to contrast those calibration techniques that are dependent on knowing the evaluation conditions with those that are robust to unseen and highly varied conditions. The latter was the goal of this work.

7.1. On the use of side information for calibration

This section commences with an ideal scenario in which evaluation and calibration training data come from the same source. To this end, we run cross-validation experiments. Specifically, the evaluation data was split into two subsets based on speaker

Table 4: Comparing the use of matched or similar data for UAC model training and DPLDA training.

Cal. Type	UAC set	DPLDA set	Avg. Cllr	Avg. Closs	Avg. EER
Global	-	matched	.243	.048	4.49%
DPLDA	matched	matched	.225	.027	4.85%
Global	-	similar	.511	.205	4.60%
DPLDA	similar	matched	.243	.041	4.55%
DPLDA	matched	similar	.502	.080	10.59%
DPLDA	matched	similar	8.17	.101	26.51%

label. Cross-evaluations were conducted in which subset A was used to train the calibration models for the evaluation of subset B and vice versa before calibrated scores were pooled and metrics were evaluated. This process is depicted in Figure 2.
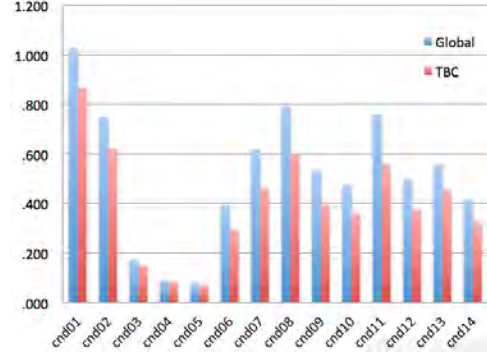
The use of matched data in this section intends to illustrate of the full potential of DPLDA by ensuring accurate extraction of side information and allowing the determination of each SI classs effectiveness prior to evaluation on mismatched calibration data in the following sections. Global calibration is compared to DPLDA with side information.

Initial experiments were intended to determine the utility of SI and DPLDA at mitigating miscalibration in the evaluation dataset. Table 3 details the different classes of SI evaluated and the relative improvement in Cllr that each provided over global calibration when averaged across the 14 conditions. The combination of both language and channel SI was found to be most effective. The last row in the table indicates the gain that could be achieved with perfect SI by calibrating each condition using data specific to that condition (i.e., 14 calibration models were used). These results indicate that, in the matched scenario of calibration, SI and DPLDA calibration is very effective in mitigating mis-calibration.
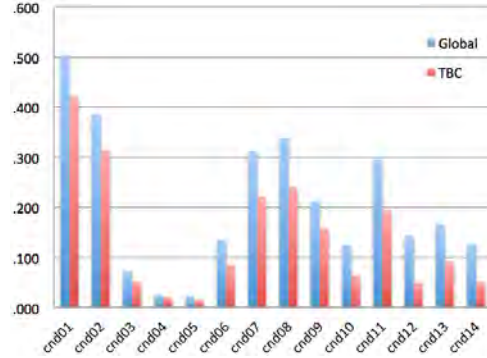
7.2. Calibration using unseen but matched data

The matched data was collected to include conditions that were similar (based on language and channel labels) to those of the evaluation dataset based on language and channel labels. In this section, we analyzed the impact of using this external data instead of the highly matched evaluation data for training the SI extractor, the DPLDA model or both components. Table 4 details results from these comparisons, and shows that the use of SRE data for the SI extractor still finds DPLDA effective as long as the DPLDA training data is well matched (evaluation data). Whenever matched data is used for DPLDA training, calibration and identification performance decrease. This is particularly the case when matched is used for training of both components.

Several conclusions can be drawn from these results. The calibration performance is very sensitive to the data used to train DPLDA. Despite the selection of SRE data that contain conditions similar to the evaluation data, the SRE data does not represent non-English cross-channel trials or cross-language trials, and does not contain all of the languages represented in the evaluation data. Given that global calibration with matched data was preferred over DPLDA when matched data was used for both SI extractor and DPLDA training, this highlights a deficiency in DPLDA to adequately accommodate unseen conditions. At worst, one would expect performance on par with global calibration.

(a) Cllr

(b) Closs

Figure 3: Illustrating the distribution of iVectors extracted from clean LRE data and heavily degraded RATS speech data.

7.3. Trial-Based Calibration (TBC)

The proposed TBC was motivated by the approach taken by forensic experts when performing SID. Specifically, the choice of data used to calibrate a trial score is delayed until conditions of both side of the trial are known. Following the procedure detailed in Section 4, we calibrated all 2.8 million scores from the evaluation set using TBC. Figure 3 compares the Cllr and Closs for global and TB calibration for each of the 14 conditions. Significant improvements of 20% and 35% in Cllr and Closs, respectively, were observed on average across the conditions. Additionally, the average EER across conditions fell from 4.60% to 4.35%. These results indicate that TBC can more readily adapt the unseen conditions than DPLDA and provide better-calibrated scores for making identification decisions across various conditions.

7.4. Unseen, Large Variability Data

The large variability data set was collected without regard to the conditions of the evaluation data with the intention of measuring the robustness of techniques to unseen evaluation conditions. This dataset was used in the evaluation of global calibration, DPLDA calibration and TBC, with the latter two utilizing UAC vectors. The UAC classes in this instance were extracted as language (seven classes in Table 2), channel (tel,mic), and degradation (clean, noisy, reverberated) as these were the most well represented classes in the calibration dataset. Results are detailed in Table 5. Compared to global calibration, DPLDA maintains Cllr on par with global calibration for conditions 01–11 but struggles with difficult conditions of 12–14.

Table 5: Comparing the use of matched or similar data for UAC model training and DPLDA training.

Cond.	Global Cal.			DPLDA Cal.			TBC		
	Cllr	Closs	EER (%)	Cllr	Closs	EER (%)	Cllr	Closs	EER (%)
01	.70	.40	1.25	.71	.35	1.71	.19	.09	0.85
02	.50	.29	0.95	.49	.24	1.90	.13	.05	0.95
03	.12	.03	1.68	.15	.02	3.35	.09	.01	1.12
04	.07	.01	1.33	.08	.02	1.33	.08	.00	1.33
05	.06	.01	0.86	.08	.01	1.93	.06	.00	1.29
06	.28	.08	4.36	.28	.01	6.70	.19	.01	3.67
07	.41	.21	4.84	.32	.03	8.36	.21	.01	4.84
08	.54	.23	6.52	.55	.04	13.26	.30	.01	6.74
09	.36	.16	2.97	.22	.03	4.65	.17	.04	3.11
10	.35	.06	7.29	.42	.02	10.55	.29	.01	6.25
11	.52	.19	6.95	.47	.01	12.59	.31	.00	7.38
12	.37	.05	8.26	.58	.05	14.22	.41	.06	8.07
13	.43	.10	9.80	.61	.07	13.73	.44	.09	7.90
14	.32	.06	7.39	.64	.09	12.61	.35	.02	7.39
Avg.	.36	.13	4.60	.40	.07	7.63	.23	.03	4.35

Closs improves using DPLDA with the average Closs dropping from 0.13 to 0.07. This comes at a cost to accuracy with a doubling of EER in some conditions. This cost can be interpreted as the systems ability to calibrate scores well by forcing the distribution of scores into a form that improves calibration but reduces separability of target and impostor trial scores. TBC on the other hand, significantly improves calibration performance on the straightforward conditions (conditions 01 and 02), and does not reduce performance in very difficult calibration conditions. The Closs through TBC is significantly reduced from 0.13 to 0.03 over global calibration. As found in the previous section, the EER of the system also improved through calibration using TBC.
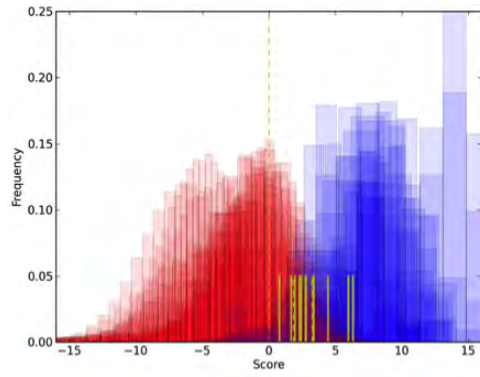
Figure 4 illustrates the threshold stability of the above techniques around the operating point with equal costs (approximate threshold of 0.0). The improved threshold stability offered by TBC over global and DPLDA calibration is clear from these plots.
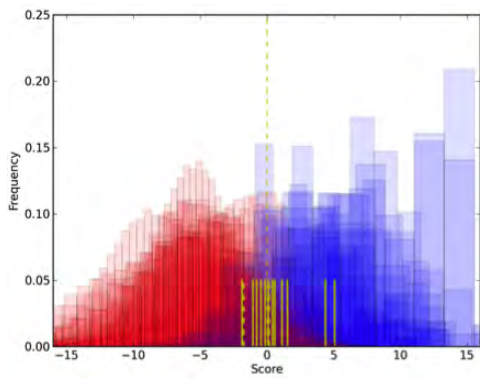
8. Conclusion

We presented a novel approach to calibration that delays the learning of calibration parameters until conditions of both sides of the trial are known. Basic approaches to calibration were shown to work well with data taken from the same corpus as the evaluation set (cross validation results on the evaluation data), however this scenario is not realistic. Using data from a data set conditioned to the characteristics of the evaluation data, this work showed that current state of the art approaches (global and DPLDA calibration) were heavily dependent on the availability of observed conditions during training. We proposed trial-based calibration (TBC) to address these issues. TBC dynamically adapts the data used to produce calibrations dependent on the conditions of the trial. That is, it removes the need to make assumptions about the evaluation scenario by waiting until trial-time where conditions can be properly accommodated. TBC reduced calibration loss and cost functions while maintaining accuracy of the SID system. The use of a diverse calibration set (the large variability dataset) was particularly beneficial in this adaptive technique when dealing with the variety of conditions in the evaluation data.
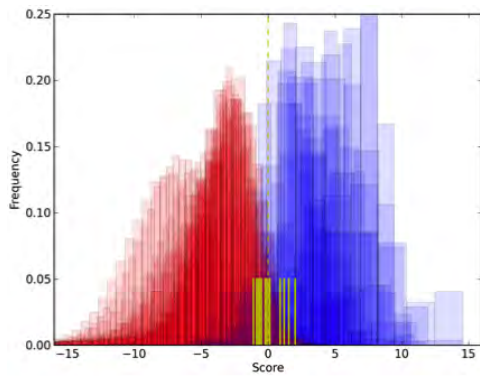
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(a) Global

(b) DPLDA

(c) TBC

Figure 4: Illustrating the distribution of trials and the thresholds for all 14 conditions at the operating point of equal cost.