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DETECTION AND CLASSIFICATION OF ACOUSTIC SCENES AND EVENTS: AN IEEE AASP CHALLENGE

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ABSTRACT

This paper describes a newly-launched public evaluation challenge on acoustic scene classification and detection of sound events within a scene. Systems dealing with such tasks are far from exhibiting human-like performance and robustness. Undermining factors are numerous: the extreme variability of sources of interest possibly interfering, the presence of complex background noise as well as room effects like reverberation. The proposed challenge is an attempt to help the research community move forward in defining and studying the aforementioned tasks. Apart from the challenge description, this paper provides an overview of systems submitted to the challenge as well as a detailed evaluation of the results achieved by those systems.

Index Terms—Computational auditory scene analysis, acoustic scene classification, acoustic event detection

1. INTRODUCTION

Over the last few years, there has been an increased interest in the speech and audio processing community in code dissemination and reproducibility of results as a means to improve the quality and relevance of published results. This can be attributed to accumulating evidence of the benefits of performing research with reproducibility in mind and making well-documented code and data publicly available [1, 2]. Public evaluation of proposed methods, especially if accompanied with open-source submissions is a key component in the move towards this reproducibility. It can serve as a reference point for the performance of proposed methods and can also be used for studying performance improvements throughout the years. Numerous initiatives have reached maturity, for example the SiSEC evaluation for signal separation [3], the MIREX competition for music information retrieval [4] and the CHiME speech separation and recognition challenge [5]. The research problems related with these evaluations are well-defined and have their own performance metrics established. However, for researchers working on modeling and classification of acoustic scenes, containing non-speech and non-music, and detecting sound events within a scene, there is not yet a coordinated established international challenge in this area, with the exception of the now discontinued CLEAR evaluations [6] funded by the CHIL project and the Multimedia Event Detection of the TRECVID video retrieval evaluations, where the focus is on audiovisual, multi-modal event detection in video recordings [7].

In this paper, we describe a newly-launched public evaluation challenge on acoustic scene classification and event detection, both for monophonic and polyphonic audio [8]. In Section 2, we present the datasets that were created for the challenge, as well as the employed evaluation metrics. Participating systems are then outlined in Section 3, and evaluation results are presented and discussed in Section 4.

2. CHALLENGE DESCRIPTION

Acoustic scene classification and detection of sound events within a scene are well defined engineering tasks that both fall under the “umbrella” of computational auditory scene analysis (CASA) [9]. The first task aims to characterize the acoustic environment of an audio stream by providing a semantic label to it [10]. The second one aims to label temporal regions within the audio, within which a specific event class is active, by estimating the start and end time of each event and if necessary (i.e. for audio extraction purposes) separate it from other overlapping events.

The present challenge consists of a set of three subtasks. The first one addresses the problem of identifying and classifying acoustic scenes or soundscapes. The other two subtasks address the problem of identifying individual sound events that are prominent in an acoustic scene: one focuses on monophonic event detection without overlapping sounds and the other focuses on polyphonic scenarios. The polyphonic case could be considered more interesting, as in realistic everyday scenarios most of the sounds that reach our ears tend to stem from a multitude of sources, but at the same time it consists of a much more challenging problem. More details about the proposed tasks along with baseline results can be found in [8, 11].

2.1. Datasets

Each of the tasks is accompanied by its own dataset. The datasets for Scene Classification (SC) consists of two equally proportioned parts each made up of ten 30 seconds recordings for each scene (class), for a total of 100 recordings per dataset. One part has been made publicly available [8] and serves as the development set for participants to investigate the performance of their system, whereas the other is kept private and used for a train/test (K-fold) evaluation. The two datasets span a pre-selected list of scene types, representing an equal balance of indoor/outdoor scenes in the London area: bus, busystreet, office, openairmarket, park, quietstreet, restaurant, supermarket, tube, tubeestation.
These recordings were made with a set of Soundman OKM II binaural microphones. These microphones imitate a pair of in-ear headphones that the user can wear for added portability and subtlety. Furthermore, the data carries also binaural information about the sound that could be utilized as cues for the sound event and scene detection or simply be ignored by adding the two channels together in order to obtain a mono recording.

The datasets for event detection were built from audio collected in office environments because of the interest of such audio to certain applications such as audio-conferencing systems etc. Two event detection tasks are proposed, a monophonic task denoted as Office Live (OL) and a polyphonic task denoted as Office Synthetic (OS). Polyphonic data for the OS task was created using a scene synthesizer, concatenating recordings of isolated events. Each dataset consists of three subsets (a training, a development and a testing subset). The training set contains instantiations of individual events in different size and absorbing quality rooms, different number of people in the room and varying noise level. Event types used were: alert (short alert (beep) sound), clearthroat (clearing throat), cough, doorslam (door slam), drawer, keyboard (keyboard clicks), keys (keys put on table), knock (door knock), laughter, mouse (mouse click), pageturn, (page turning), pendrop (pen, pencil, or marker touching table surfaces), phone, printer, speech, switch. To capture the spatial layout of the acoustic environment, recordings were made in first order B-format (4-channel), with a high-quality Soundfield SPS422B microphone system, placed in an open space in the room, with events spatially distributed around the room. Recordings were mixed down to stereo (using the common “Blumlein pair” configuration). The challenge is conducted using the stereo files, with scope to extend the challenge to full B-format in future if there is demand.

More details about the creation of the datasets, the annotation process and the audio recording process can be found in [8].

### 2.2. Evaluation Metrics

For the scene classification task, systems are evaluated with 5-fold stratified cross validation. The raw classification (identification) accuracy, standard deviation and a confusion matrix for each algorithm is computed.

For the event detection tasks, in order to provide a thorough assessment of the various systems, three types of evaluations take place, namely a frame-based, event-based, and class-wise event-based evaluation. The main metrics used for each evaluation type are the F-measure ($F$) and the acoustic event error rate ($AEER$) as described in [8]. For the event-based and class-wise event-based metrics, two types of evaluation will take place, an onset-only and an onset-offset-based evaluation.

Results to onset-based metrics (denoted without any subscript) and onset-offset-based metrics (denoted as $F_{offset}$ and $AEER_{offset}$). For a complete and analytic description of the evaluation metrics employed the reader is referred to [8, 11].

### Table 1: Summary of submitted scene classification systems.

<table>
<thead>
<tr>
<th>Participants</th>
<th>Code</th>
<th>Method</th>
<th>Lang</th>
</tr>
</thead>
<tbody>
<tr>
<td>Ch hemorrh. et al.</td>
<td>CHR</td>
<td>Various features at 2 frame sizes, classified either: (a) per-frame SVM + majority voting; (b) HMM</td>
<td>Matlab</td>
</tr>
<tr>
<td>Elizalde</td>
<td>ELF</td>
<td>Concatenation of 4 different mono mixdowns; “i-vector” analysis of MFCCs, classified by pLDA</td>
<td>Matlab</td>
</tr>
<tr>
<td>Geiger et al.</td>
<td>GSR</td>
<td>Diverse features, classified within 4-second windows using SVM, then majority voting</td>
<td>Matlab</td>
</tr>
<tr>
<td>Krijnders and ten Holt</td>
<td>KH</td>
<td>&quot;Cochleogram&quot; representation, analysed for tonelikeness in each t-f bin, classified by SVM</td>
<td>Python</td>
</tr>
<tr>
<td>Li et al.</td>
<td>LTT</td>
<td>Wavelets, MFCCs and others, classified in 5-second windows by treebagger, majority voting</td>
<td>Matlab</td>
</tr>
<tr>
<td>Nam et al.</td>
<td>NHL</td>
<td>Feature learning by sparse RBM, then event detection and max-pooling, classified by SVM</td>
<td>Matlab</td>
</tr>
<tr>
<td>Nogueira et al.</td>
<td>NR1</td>
<td>MFCCs + MFCC temporal modulations + event density estimation + binaural modelling features, feature selection, classified by SVM</td>
<td>Matlab</td>
</tr>
<tr>
<td>Oliveti</td>
<td>OE</td>
<td>Normalised compression distance (Vorbis), Euclidean embedding, classified by Random Forest</td>
<td>Python</td>
</tr>
<tr>
<td>Patil and Elhilali</td>
<td>PE</td>
<td>Auditory representation analysed for spectrotemporal modulations, classified within one-second windows using SVM, then weighted combination of decision probabilities</td>
<td>Matlab</td>
</tr>
<tr>
<td>Rakotomamonjy and Gasso</td>
<td>RG</td>
<td>Computer vision features (histogram of oriented gradient) applied to constant-Q spectrogram, classified by SVM</td>
<td>Matlab</td>
</tr>
<tr>
<td>Roma et al.</td>
<td>RNH</td>
<td>Recurrence Quantification Analysis applied to MFCC time-series, classified by SVM</td>
<td>Matlab</td>
</tr>
<tr>
<td>Baseline</td>
<td></td>
<td>MFCCs, classified with a bag-of-frames approach</td>
<td>Python</td>
</tr>
</tbody>
</table>

### Table 2: Summary of submitted event detection systems.

<table>
<thead>
<tr>
<th>Participants</th>
<th>Code</th>
<th>Method</th>
<th>Lang</th>
</tr>
</thead>
<tbody>
<tr>
<td>Chauhan et al.</td>
<td>CPS</td>
<td>Feature extraction - Segmentation - Likelihood ratio test classification</td>
<td>Matlab</td>
</tr>
<tr>
<td>Diment et al.</td>
<td>DHV</td>
<td>MFCCs (features) - HMMs (detection)</td>
<td>Matlab</td>
</tr>
<tr>
<td>Gemmeke et al.</td>
<td>GVV</td>
<td>NMF (detection) - HMMs (postprocessing)</td>
<td>Matlab</td>
</tr>
<tr>
<td>Niesen et al.</td>
<td>NVM</td>
<td>Hierarchical HMMs + Random Forests (classification) - Meta-classification</td>
<td>Matlab</td>
</tr>
<tr>
<td>Nogueira et al.</td>
<td>NR2</td>
<td>MFCCs (features) - SVMs (classification)</td>
<td>Matlab</td>
</tr>
<tr>
<td>Schröder et al.</td>
<td>SCS</td>
<td>Gabor filterbank features - HMMs (classification)</td>
<td>Matlab</td>
</tr>
<tr>
<td>Vugem et al.</td>
<td>VVK</td>
<td>MFCCs (features) - GMMs (detection)</td>
<td>Matlab</td>
</tr>
<tr>
<td>Baseline</td>
<td></td>
<td>NMF with pre-extracted bases (detection)</td>
<td>Matlab</td>
</tr>
</tbody>
</table>
3. SUBMITTED SYSTEMS

Overall, 11 systems were submitted to the SC task, 7 systems were submitted to the OL task, and 3 systems to the OS task. Variants for each system were allowed that increased the total number and variety considerably. The systems submitted for the scene classification and event detection tasks are listed in Tables 1 and 2, respectively, along with a short description of each system and the programming language in which it was written.

Apart from the submitted systems, performance on the test sets is also reported for baseline systems for the two tasks. These systems were made publicly available as open source software [11].

4. CHALLENGE RESULTS

Results were computed by running all the submitted systems on the held-back testing datasets and computing the metrics as in Sec. 2.2. Figure 1 shows the overall performance of submitted systems for the scene classification task. Most systems were able to outperform the baseline system, and some matched or even slightly excelled the mean accuracy we found in human listeners (71%; results in preparation). The strongest performers are notably diverse in their choice of features and their use of temporal information, though often using SVMs for classification. Two submissions achieved good results on the development data but not on our held-out test data.

Table 3 shows a confusion matrix for the scene labels as round percentages of the sum of all confusion matrices for all submissions. Confusions are mostly concentrated over classes that share some acoustical properties such as "park/quietstreet and tube/tubestation."

For the event detection OL and OS tasks, results are summarized in Tables 4 and 5, respectively. The baseline was outperformed by most systems for these tasks too. The best performance for the OL task using all types of metrics is achieved by the SCS submission, which used a Gabor filterbank feature extraction step with a 2-layer hidden Markov models (HMMs) for classifying events, followed by the NVM submission, which used a meta-classifier combining hierarchical HMMs and random forests. For the OS task, the best performance in terms of F-measure is achieved by the DHV system, which used an iterative scheme with HMMs. It should also be noted that submitted systems performed better with lower polyphony, with the exception of the DHV system, which had better performance with higher polyphony levels. As expected, the onset-offset evaluation produced worse results compared to onset-only evaluation for both tasks, although the performance difference is rather small. This may be explained by the percussive nature of most events.

The challenge website [8] gives detailed system descriptions and extensive results, analytic breakdown of performance per system, as well as further error analysis.

5. CONCLUSIONS

In this paper we presented a challenge on the detection and classification of acoustic scenes and events. We ran a scene classification (SC) challenge, and two event detection and classification challenges: office live (OL) and office synthetic (OS). Our goal was to provide a focus of attention for the scientific community in developing systems for CASA that will encourage sharing of ideas and improve the state of the art, potentially leading to the development of systems that achieve a performance close to that of humans.

The results enable us to draw some interesting conclusions about the different problems. For scene classification, we found that although simple systems can do relatively well, the improvement that more complex systems achieve can bring performance to the levels achieved by human listeners. For event detection, which is a more challenging task, performance is much worse although we have not performed a direct comparison with human listeners at present. For the monophonic case, systems are able to achieve satisfactory performance with scope for improvement. For the polyphonic case, the task of recognising individual potentially overlapping sounds becomes significantly challenging and the performance of systems that are even prepared to deal with polyphonic content.
falls dramatically. More details for all the submitted systems can be found on the challenge website in [8].

At this point, we have just completed running the challenge. For future work, we will consider producing a detailed performance evaluation, creating a code repository, releasing test sets, doing a B-format challenge, running the challenge again or doing a challenge evaluation campaign (SiSEC2011), in "Latent Variable Analysis: Principles, algorithms, and applications," in IEEE Int. Conf. Acoustics, Speech and Signal Processing, 2007, pp. 1273–1276.

6. ACKNOWLEDGMENTS

The authors would like to thank all of the contributors, for taking part in the challenge and participating in email discussions as we were developing the challenge, and Daniele Barchiesi, for his contribution in running the evaluation of the submitted systems and providing useful feedback.

7. REFERENCES