Spatiotemporal Analysis of Urban Noise Data

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1 Introduction

Noise Pollution is a grievous problem which needs to be tackled, as it's a growing concern for many urban residents. Sounds that are not particularly loud but are nonetheless undesirable and uncontrollable can have serious implications for the listener, particularly if they occur over a long period of time. The noise becomes much more upsetting if the source of noise is an agent or agency that has shown little concern for the individual who is suffering from the noise's effects and, as a result, has done nothing to reduce the noise. Noise is not only inconvenient and annoying, but it has also been proven to be a health hazard, many have reported that they suffered with behavioral and emotional consequences, such as difficulty in sleeping, relaxing and feeling annoyed, angry or upset [3,4,7] and When intrusive noises continue, the body responds physiologically, and there is a risk of irreversible bodily damage damage to the circulatory, cardiovascular, and gastrointestinal systems over time.

In order to mitigate this problem, there is a need to understand sound event detection. Sound event detection is defined as recognition of individual sound events in audio, e.g., "dog barking, engine exhaust noise" requiring estimation of onset and offset for distinct sound for sound event detection and identification of sound. Applications for sound event detection can found in areas of Healthcare, security, audio and video-based indexing and retrieval. Sound class classification is usually approached as a supervised learning, with sound classes defined beforehand, we have taken a labeled Spatial and Temporal recording data which comprises of 3068 labeled 10 sec recordings from the Sounds of New York City (SONYC) acoustic network (An acoustic network is a method of positioning equipment using sound waves). Using this data, we plan to develop an application where the authorities or the user can narrow down the sounds generated at any particular location. We also aim to address the Mismatch of the testing data in this dataset.

This can be done by applying machine learning algorithm on any dataset and integrating it with sensors, data analytics for the development of machine learning systems for real world urban noise monitoring. The model build from the data could be used on neighborhoods to better understand noise in that location and help the authorities mitigate the issue. The many challenges for building the model would be like to separate the sound sources of interest, identifying the similar sounds compared to the other data in the dataset, identifying the main source for generating the sound among others.

2 RELATED WORK

Collecting audio data and annotating the soundscapes are an important part of acoustic research especially in the situations with high variability in different locations.

In the paper [6] they have discussed about soundscapes and different ways to annotate audio data using crowdsourcing. There are basically two ways to achieve this one of them being waveform and the other one being spectrogram visualization. Certain annotation trials were done using the two techniques and interesting results were drawn from the experiments. People using the spectrogram visualization technique were

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able to produce high quality and precision annotations than waveform visualization.

According to [2] a supervised learning methodology is applied to real-life high quality recordings of 3-5 minutes with very little noise of 15 different acoustic scenes (lakeside beach, bus, cafe/restaurant, car, city center, forest path, grocery store, home, library, metro station, office, urban park, residential area, train, and tram) and two common environment areas (outdoor - residential areas and indoor - home). A Mel frequency Cepstral coefficient (MFCC) and Gaussian mixture model (GMM) was trained using expectation maximization algorithm. The overall accuracy achieved by the model is 72.5 % ranging from 13.9% for parks to 98.6% for office spaces. With our system we plan to include cross-validation to train our model so that there is no data contamination between train and test sets.

According to the work done in [5] Audio annotation is critical for creating machine-listening systems, but there is little research on how to get accurate and timely crowd sourcing audio annotations. They aimed to quantify the reliability/redundancy trade-off in crowd sourced soundscape annotation, look at how visualizations affect accuracy and efficiency, and characterize how performance varies as a function of audio characteristics.

Our application will be based on the research paper's data [5], which presents us the process to collect acoustic data. SONYC has developed an acoustic sensor with high quality and low production cost to monitor the noise pollution levels across the city in neighborhoods like Manhattan, Brooklyn and Queens. The collected data was then annotated using a campaign on Zooniverse. The sensors follow DCASE (Detection and classification of acoustic scenes and events) to eliminate discrepancy. There are various other datasets like UrbanSound, UrbanSound8k that address this particular problem but have limited spacial and temporal data points. A VGGish model has been developed and trained using stochastic gradient descent to minimize cross-entropy loss. To eliminate over-fitting early stopping on validation set has been implemented. Two models were trained on course-level and fine-level tags. The overall AUPRC achieved by this model is 0.62 and 0.76 on different level classes, which performed poorly on music and nonmachinery impact sounds. We plan to use this model and analyze the mismatches caused by the prediction on actual test data and find out the causes which led to this mismatch.

Over the years there has been a lot of research on annotation of audio data in different scenarios and predicting the source of noise in big cities. We will be extending this to analyze the mismatch of such predictions by leveraging machine learning metrics and coming up with our own model to predict the results with higher accuracy. A big part of our research will also be visualizing the locations of these mismatches. However most of our work focuses on analyzing and developing models for the (SONYC-UST) dataset [1].

3 DATA DESCRIPTION

In order to build a system which would visualize the various sound points in and around a particular geographical location and also allow us to analyse the mismatches between the test and machine data, we need to have a dataset which has a diverse distribution of labeled sounds with spatial and temporal attributes. For which we have taken a dataset containing a training subset (13538 recordings from 35 sensors), validation subset (4308 recordings from 9 sensors), and a test subset (669 recordings from 48 sensors). Each recording has been annotated using a set of 23 "sound tags" like "engine presence, machinery presence, non-machinery-impact presence, dog-barking-whining presence, music presence etc." [5].

The training, validation, and test subsets of the annotation data are contained in annotations.csv (see Figure 1). Each row in the file represents one multi-label annotation of a recording—it might be a single citizen science volunteer's annotation, a single SONYC team member's annotation, or the SONYC team's agreed-upon ground truth (for more information, see the annotator id column description). The audio files used were recorded using the SONYC acoustic sensor network for monitoring urban noise pollution. In New York City, over 60 distinct sensors have been placed accumulating the equivalent of more than 50 years of audio data, of which a small fraction was used. The data was sampled by picking the closest neighbors based on VGGish qualities of recordings with recognized classes of interest. All of the recordings are 10 seconds long and were made with the same microphones and gain settings.

The sensors in the test set will not disjoint from the training and validation subsets, but the test recordings are displaced in time, occurring after any of the recordings in the training and validation subset. We plan to use the test data to find out the aggregate of mismatch by using a Multi Label classification Machine Learning Model.

In our work we would utilize the latitude and longitude columns to spatially locate the sound origin, the year, week, day, hour columns to precisely point the time and the engine, machinery-impact, non-machinery-impact, powered-saw etc. columns to get the sound type and analyze and visualize the data.

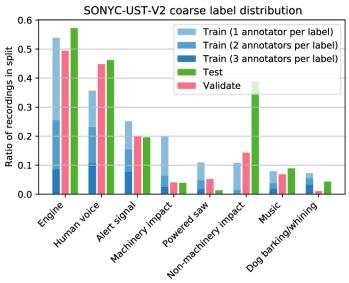


Fig. 1. Distribution of various sound tags in test, train, validate sets of dataset

4 PROPOSAL

We propose to build a tool which would visualize the spatiotemporal values of the dataset, visuliaze the output of Machine Learning model and also visualize the points where the mismatches of testing and machine data occur based on the results of the Machine Learning model. We will be using Multi Label Classification Machine Learning Algorithms to to classify the audio files into various categorical sounds. Based on the results of ML model and mismatch plot, we would dwell into understanding the mismatches which occur between the annotated and machine predicted data and analyze the causes behind it.

The technologies which we plan to use in order to build this tool would be ReactJS on the front-end and Django on the back-end. We are using ReactJS because of its component based architecture and its rich support with D3.js, Open Layers. Similarly, Django allows us run various python libraries to process the data and creation of REST API's (representational state transfer application programming interface) to communicate with the front-end.

We have divided the process of building this tool into four stages(see Figure 2) namely Data Preprocessing state wherein we process the

audio files,Modeling stage where in Applying the Classification model on the dataset and creating REST API's to interact with the frontend, Visualization stage where in we visualize the results of machine learning model on the front-end,Component building stage where we build components which allows the user to pick spatial and temporal values of various sounds and displaying the results on a map.

The built tool would be useful to authorities who monitor sounds around the city would allow them to pinpoint the location of sound origin.

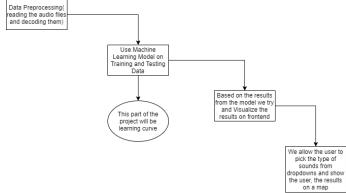


Fig. 2. Work flow diagram

5 TIMELINE

Week	Task
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Week 1	First week is to Analyze and Preprocess the dataset.
Week 2	Parsing the dataset and Extracting features from Audio Files.
Week 3	Analyses of the extracted features.

Week 4	Construct and apply the machine learning model on the given data.
Week 5	Analyze and Infer results of the model.
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Week 6	Build a Back-end system and develop API's which would serve with spatial and temporal responses to visualize on Front-end.
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Week 7	David of ADV single Park and
week /	Develop API's in the Back-end
Week 8	Develop Front-end component - Spatial-Temporal analysis of the dataset based on back-end responses.
Week 9	Develop Frontend component - Which would depict and address the mismatch problem of the dataset.
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Week 10	Handling the edge cases of the system and improving on the system.

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