

## The Sound Classification System Based on Neural Networks with Attention Mechanism and Autoencoders

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| ARTICLE INFO   | ABSTRACT  |
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| <p>Published Online:<br/>02 July 2020</p> <p>Corresponding Author:<br/><b>Anastasiia Kryvokhata</b><br/>Zaporizhzhya National<br/>University, Zhukovsky str., 66,<br/>Zaporizhzhya, 69600, Ukraine<br/>+380972100376</p> | <p>This paper discusses the models and methods of machine learning methods used in environmental sound classification systems. A sound classification subsystem could be implemented in the various Smart City, Smart Farming systems, healthcare etc. An automated sound classification system could be decomposed into following subsystems: the audio representation, the features extraction, the classifier and the accuracy estimation. This paper deals with a convolutional neural network with attention mechanism for sound classification and autoencoder for data augmentation. The objective of the paper is to develop the optimal sound classifier for datasets with lack of observations.</p> |
| <p><b>KEYWORDS:</b> Sound Classification, Convolutional Neural Network, Autoencoder, Attention.</p>  |   |

### I. INTRODUCTION

Machine hearing systems [1] are quite useful in such areas as audio surveillance, automatic medicine auscultation, acoustic scene detection in urban environments etc. The main requirements for such systems are real time classification with high accuracy and ability to generalization on unseen data. Environmental sound classification differs from music or speech recognition by higher data variability and lack of frequent pattern repetition.

There are different approaches to environmental sound classification but most suitable method here is neural network classifier. The benefits of using neural networks are as follows. Generalization to new data allows classifying new sounds using a pre-trained model. Deep learning methods allow extracting features from raw sound, so there is no need in additional sound preprocessing modules. However, a large amount of training data is required for neural networks training that is disadvantage. This can be overcome by using autoencoders for data augmentation. There are different approaches for accelerating learning process and improving accuracy: meta-learning, evolutionary algorithms etc. Particularly, attention mechanism for neural networks has been used successfully in recent articles to improve the classifier accuracy. Therefore, our goal is to develop environment sound classification system using autoencoder for data

augmentation and deep convolutional neural network with attention mechanism as classifier.

### II. THE AIM AND OBJECTIVES OF THE STUDY

There are several essential steps in sound classification systems, we consider following subsystems: audio data preprocessing, features extraction, classification, and accuracy estimation. The audio data preprocessing stage implies different approaches such as segmentation, filtering, transformations. Equal sound files could be obtain by segmentation of the raw signal into shorter chunks using some windowing process. Thus, the original acoustic signal is converted into the frames of a fixed length. Trailing silence frames from an audio signal could be done using sound energy filters.

The feature extraction stage includes methods for different representations of the initial digital signal. This stage exploits special features such as Mel Frequency Cepstral Coefficients (MFCC), Gammatone Mel Frequency Cepstral Coefficients (GFCC), chromagrams and spectrograms. The aim of this stage is to find better signal characteristic for further classification.

The State-of-the-art methods for sound classification are neural networks and other machine learning methods like K-means, support vector machine (SVM), decision trees etc [2-4]. The classifiers based on convolutional or recurrent neural networks are quite effective in these problems. There are

different approaches for neural networks fine tuning, for instance, meta-learning, genetic algorithms and special layers like attention mechanism.

The aim of this paper is to develop a proof-of-concept system for the environmental sound classification based on convolutional neural networks and autoencoders. We suggest that autoencoders could be used for data augmentation and convolutional model could be improved by attention layer.

In order to reach the mentioned aim, the following objectives were formulated for the study:

- to review state-of-the-art environmental sound classification systems;
- to develop autoencoder model for data augmentation;
- to develop classification system using convolutional neural networks with attention layer;
- to outline a direction for further development of machine hearing systems.

We going to use ESC-50 dataset for training and testing classification model. This dataset consists of 5-second-long recordings organized into 50 semantical classes with 40 examples per class [5].

### III. LITERATURE REVIEW

There are numerous survey papers have published recently on the topic of sound classification, for instance, articles [2,

3] provide a description of the components of an automatic sound classification system, which contains preprocessing modules, feature extraction, training algorithm and evaluation module.

The signal feature extraction methods are discussed in details [2]. The physical properties of the signals and the characteristics of the human perception of sounds are two basic category here. Feature extraction methods represent the acoustic signal in the time, frequency, cepstral and wavelet domains.

The analysis of general approaches to automatic music classification by genre is described in [3, 6].

The methods of neural networks application in the feature extraction and the classification are given in [1, 6, 7]. Deep convolutional neural networks are the main classifier in this area, which is explained by the complex layered architecture with several layer types. The disadvantages of this approach include the complexity of neural networks fine tuning, demand for computing resources and large amount of observations in the learning datasets.

Attention mechanism is described in [8-10]. This approach demonstrates high accuracy and become an essential part of neural network classification architectures.

### IV. METHODOLOGY

The workflow of automated environmental sound classification system has shown in the figure 1.

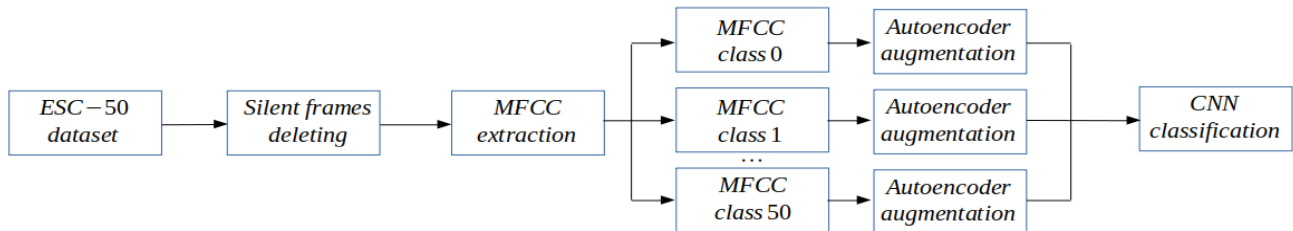


Figure 1: Sound classification system workflow

| Layer (type)                 | Output Shape         | Param # |                              |                          |
|------------------------------|----------------------|---------|------------------------------|--------------------------|
| input_36 (InputLayer)        | [(None, 20, 431, 1)] | 0       | leaky_re_lu_284 (LeakyReLU)  | (None, 1, 6, 32) 0       |
| conv2d_315 (Conv2D)          | (None, 20, 431, 32)  | 608     | up_sampling2d_140 (UpSamplin | (None, 2, 24, 32) 0      |
| leaky_re_lu_280 (LeakyReLU)  | (None, 20, 431, 32)  | 0       | conv2d_320 (Conv2D)          | (None, 2, 24, 32) 18464  |
| average_pooling2d_140 (Avera | (None, 10, 108, 32)  | 0       | leaky_re_lu_285 (LeakyReLU)  | (None, 2, 24, 32) 0      |
| conv2d_316 (Conv2D)          | (None, 10, 108, 32)  | 18464   | up_sampling2d_141 (UpSamplin | (None, 4, 72, 32) 0      |
| leaky_re_lu_281 (LeakyReLU)  | (None, 10, 108, 32)  | 0       | conv2d_321 (Conv2D)          | (None, 4, 72, 32) 18464  |
| average_pooling2d_141 (Avera | (None, 5, 27, 32)    | 0       | leaky_re_lu_286 (LeakyReLU)  | (None, 4, 72, 32) 0      |
| conv2d_317 (Conv2D)          | (None, 5, 27, 32)    | 18464   | up_sampling2d_142 (UpSamplin | (None, 8, 216, 32) 0     |
| leaky_re_lu_282 (LeakyReLU)  | (None, 5, 27, 32)    | 0       | conv2d_322 (Conv2D)          | (None, 8, 216, 32) 18464 |
| average_pooling2d_142 (Avera | (None, 2, 6, 32)     | 0       | leaky_re_lu_287 (LeakyReLU)  | (None, 8, 216, 32) 0     |
| conv2d_318 (Conv2D)          | (None, 2, 6, 32)     | 18464   | up_sampling2d_143 (UpSamplin | (None, 24, 432, 32) 0    |
| leaky_re_lu_283 (LeakyReLU)  | (None, 2, 6, 32)     | 0       | conv2d_323 (Conv2D)          | (None, 20, 431, 1) 321   |
| average_pooling2d_143 (Avera | (None, 1, 6, 32)     | 0       |                              |                          |
| conv2d_319 (Conv2D)          | (None, 1, 6, 32)     | 18464   | Total params: 130,177        |                          |
|                              |                      |         | Trainable params: 130,177    |                          |
|                              |                      |         | Non-trainable params: 0      |                          |

Figure 2: Autoencoder architecture

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We use MFCC as features for further classification but there are only 40 5-second-long observations per class, so some method of data augmentation is required here. Thus, for 2000 initial observations we obtain 2000x20x431 matrix representation for 20 Mel Frequency Cepstral Coefficients per each sound file. Autoencoder could be used for generating new sounds for each class. The convolutional autoencoder architecture has shown in the figure 2.

Keras library is used for neural network implementation. We use Conv2D layers for convolutions and AveragePooling layers for dimension reduction and Leak ReLu is use as activation function.

The loss functions for different autoencoders classes change during learning process as has shown in the figure 3.

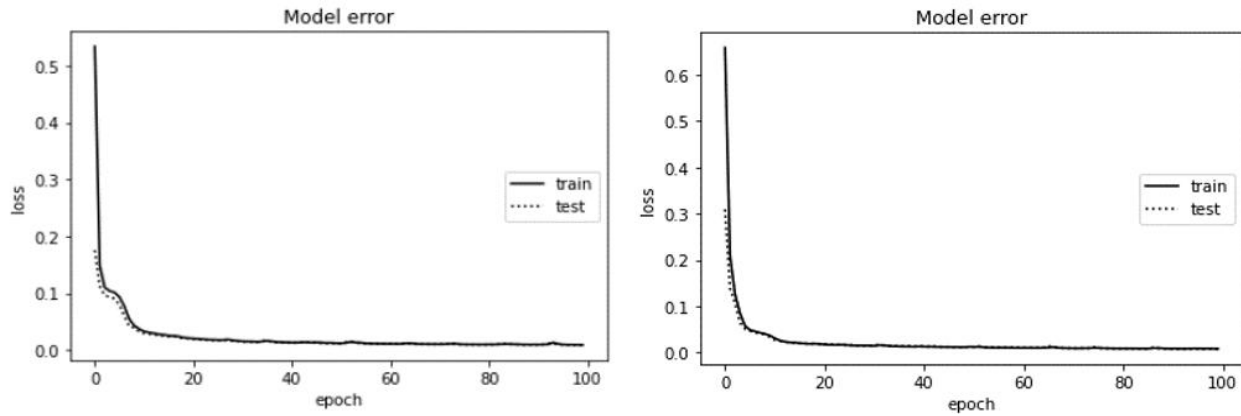


Figure 3: Autoencoders loss function

```

1 reg = 0.001
2 n_classes = 50
3 batch_size = 512
4 kernel_initializer= 'he_uniform'
5 bias_initializer='zeros'
6 kernel_regularizer=regularizers.l2(reg)
7 activation = "relu"
8 data_rows = 20
9 data_cols = 431
10 n_epochs = 100
11 input_shape = (data_rows, data_cols, 1)
12 print(input_shape)
13 n_units = data_rows*data_cols
14 alpha_zero = 0.0001
15 print('Building model...')
16 inp = Input(shape = (input_shape))
17 x = Conv2D(32, (3, 5), padding="same",
18         data_format="channels_last",kernel_initializer=kernel_initializer,
19         bias_initializer=bias_initializer, kernel_regularizer=kernel_regularizer)(inp)
20 x = Activation(activation)(x)
21 x = MaxPooling2D(pool_size=(2,2))(x)
22 x = Conv2D(64, (3, 1), padding="same",
23         data_format="channels_last",kernel_initializer=kernel_initializer,
24         bias_initializer=bias_initializer, kernel_regularizer=kernel_regularizer)(x)
25 x = Activation(activation)(x)
26 x = MaxPooling2D(pool_size=(2,1))(x)
27 x = Conv2D(256, (1, 5), padding="same",
28         data_format="channels_last",kernel_initializer=kernel_initializer,
29         bias_initializer=bias_initializer, kernel_regularizer=kernel_regularizer)(x)
30 x = Activation(activation)(x)
31 x = MaxPooling2D(pool_size=(1,2))(x)
32 x = Conv2D(256, (3, 3), padding="same",
33         data_format="channels_last",kernel_initializer=kernel_initializer,
34         bias_initializer=bias_initializer, kernel_regularizer=kernel_regularizer)(x)
35 x = Activation(activation)(x)
36 x = MaxPooling2D(pool_size=(1,2))(x)
37
38 x = Conv2D(256, (3, 3), padding="same",
39         data_format="channels_last",kernel_initializer=kernel_initializer,
40         bias_initializer=bias_initializer, kernel_regularizer=kernel_regularizer)(x)
41 x = Activation(activation)(x)
42 x = MaxPooling2D(pool_size=(1,2))(x)
43 x = Reshape((130, 256))(x)
44 x = Bidirectional(GRU(256, activation='tanh', dropout=0.5, return_sequences=True))(x)
45 x = Bidirectional(GRU(256, activation='tanh', dropout=0.6, return_sequences=True))(x)
46 att = attention_3d_block(x)
47 output = Dropout(0.6)(att)
48 prediction = Dense(n_classes,kernel_initializer=kernel_initializer,bias_initializer=bias_initializer)(output)
49 prediction = Activation('sigmoid')(prediction)
50
51 model = models.Model(inputs = inp, outputs = prediction)

```

Figure 4: Convolutional neural network implementation

The program implementation of classifier has shown in the figure 4. The neural network hyperparameters like number of epochs, batch size, learning rate are determine empirically after numerical experiments.

The data preprocessing stage includes trailing silence from an audio signal and calculating of mel frequency cepstral coefficient (MFCC) for the giving sound files. This approach allows to unify and simplify the sound files presentation in the memory. Further we feed MFCC arrays to the convolutional neural net.

V. NUMERICAL RESULTS

The confusion matrix is most appropriate metric for model testing in sound classification. The testing dataset includes 8 raw sound files per each class from ESC-50 dataset. The resulting confusion matrix has been shown in the figure 5. At the matrix each row and column means following categories:

- 0: {'dog'}, 1: {'rooster'}, 2: {'pig'}, 3: {'cow'}, 4: {'frog'}, 5: {'cat'}, 6: {'hen'}, 7: {'insects'}, 8: {'sheep'}, 9: {'crow'}, 10: {'rain'}, 11: {'sea\_waves'}, 12: {'crackling\_fire'}, 13: {'crickets'}, 14: {'chirping\_birds'},

- 15: {'water\_drops'}, 16: {'wind'}, 17: {'pouring\_water'}, 18: {'toilet\_flush'}, 19: {'thunderstorm'},
- 20: {'crying\_baby'}, 21: {'sneezing'}, 22: {'clapping'}, 23: {'breathing'}, 24: {'coughing'}, 25: {'footsteps'},
- 26: {'laughing'}, 27: {'brushing\_teeth'}, 28: {'snoring'}, 29: {'drinking\_sipping'}, 30: {'door\_wood\_knock'},
- 31: {'mouse\_click'}, 32: {'keyboard\_typing'}, 33: {'door\_wood\_creaks'}, 34: {'can\_opening'},
- 35: {'washing\_machine'}, 36: {'vacuum\_cleaner'}, 37: {'clock\_alarm'}, 38: {'clock\_tick'}, 39: {'glass\_breaking'},
- 40: {'helicopter'}, 41: {'chainsaw'}, 42: {'siren'}, 43: {'car\_horn'}, 44: {'engine'}, 45: {'train'},
- 46: {'church\_bells'}, 47: {'airplane'}, 48: {'fireworks'}, 49: {'hand\_saw'}.

From this confusion matrix we can see that approach with autoencoder augmentation can lead to quite accurate classification. The maximum row element in the matrix is on the diagonal for most classes. Thus, there are 253 observations predicted as true positive from total 400, therefore, there is room for improvement the accuracy.

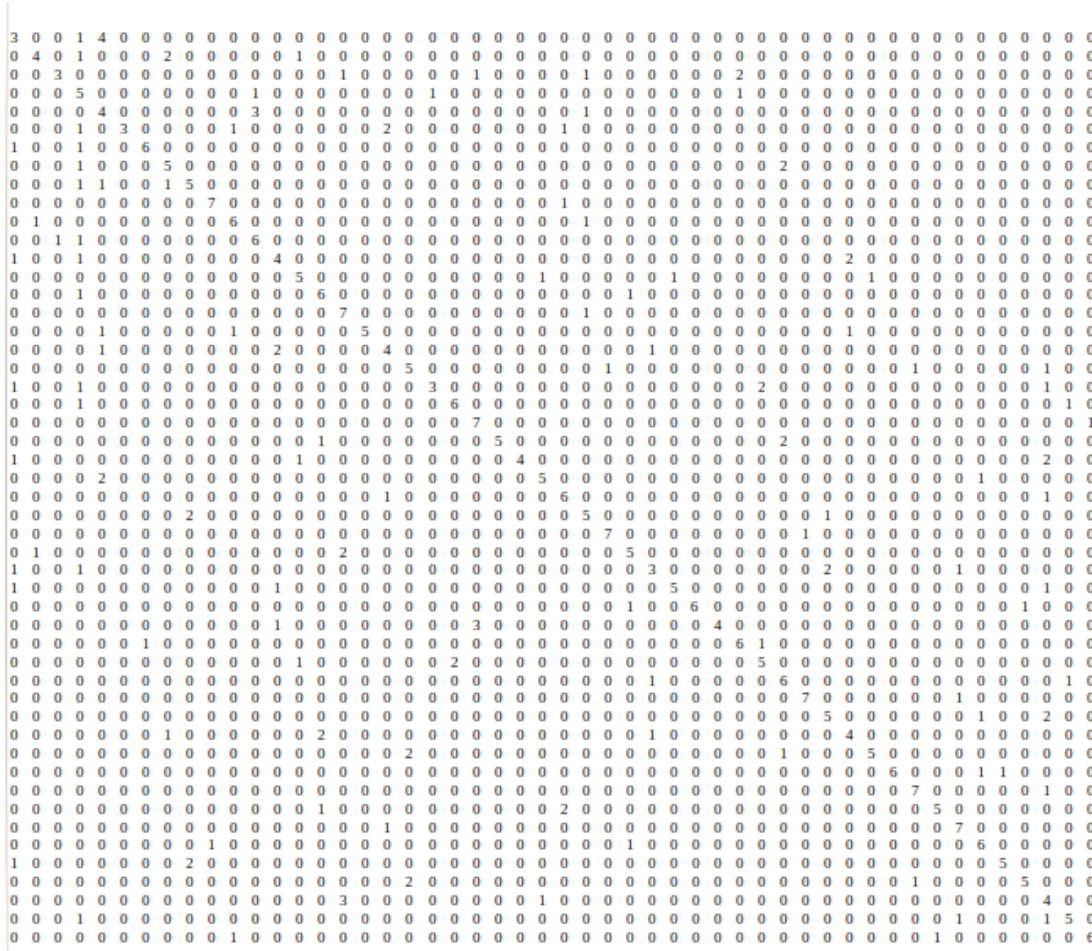


Figure 5: Sound classification confusion matrix

VI. CONCLUSION

A convolutional neural network with attention layer could be an effective classifier even in the case of train data lack. Convolutional auto encoder was used to generate new observations.

The basic distinction of this paper from papers [9] is that we tried to use a data set with only 40 observations per class. Obtained accuracy in 63% allows to continue research in this area.

The prospect of further research is related to the extension of the considered approach with meta-learning methods, for instance MAML or Reptile.

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