Knock-Knock: Acoustic Object Recognition by using Stacked Denoising Autoencoders

Shan Luo, Leqi Zhu, Kaspar Althoefer, Hongbin Liu

Abstract—This paper presents a successful application of deep learning for object recognition based on acoustic data. It can restrict capability of the representation to serve different applications and may only capture insignificant characteristics for a task when using handcrafted features. In contrast, there is no need to define the feature representation format when using multilayer/deep architecture methods and features can be learned from raw sensor data without defining discriminative characteristics a-priori. In this paper, stacked denoising autoencoders are applied to train a deep learning model. Thirty different objects were classified in our experiment and each object was knocked 120 times by a marker pen to obtain the auditory data. By employing the proposed deep learning framework, a high accuracy of 91.50% was achieved. The traditional method using handcrafted features with a shallow classifier was taken as a benchmark and the attained recognition rate was only 58.22%. Interestingly, a recognition rate of 82.00% was achieved when using a shallow classifier with raw acoustic data as input. Nevertheless, the time taken for classifying one object using deep learning was far less (6.77 faster) than utilizing this method. It was also explored how different model parameters in deep architecture would affect the recognition performance.

Keywords — Object recognition, deep networks, acoustic data analysis.

I. INTRODUCTION

Future intelligent robots are envisioned to be endowed with perceptive capabilities to see, touch and hear what is happening in the ambient world. It enables robots perform various tasks and object recognition is among the most common and significant ones. To perform this task, many types of sensors can be utilized and each kind of sensor offers a different view of objects. One of the richest and most widely used sensors is the camera as much information can be acquired from one single image. Because of this, vision has attracted considerable attention in object recognition by classifying the color [1], texture [2], [3], surface reflectance [4] and appearance [5], [6]. But vision is heavily dependent on the surrounding environments and would fail due to the variance of poses, illumination changes or occlusion by other objects. Another not as powerful but also very rich sensing modality is the sense of touch. With the use of force/tactile sensors, the object properties can be revealed by accessing the information of hardness/softness [7], elasticity [8], thermal cues [9], surface texture [10] and surface friction properties [11]. However, the tactile object recognition needs the direct contact with the objects and there will be a risk of damage when handling objects of hazardous materials. As an alternative acoustic data can be acquired by sensors like microphones to recognize the objects. It can allow the robot to work in safer conditions. In addition, sound signals generated by striking an object can expose the intrinsic properties of objects such as elasticity and internal friction [12]. The elasticity of an object is directly related to the speed of sound waves in the object and therefore influences the frequency of the sound. The internal friction, or dampness, determines how the generated sounds decay over time [12] and provides shape-variant acoustic features for object classification [13].

To date, however, audition has been largely neglected relative to vision and tactile sensation in the application of object recognition. One of the most dominant factors is that the auditory data is more abstract compared to visual images and force/tactile data. In the traditional acoustic based recognition, the task is achieved by using handcrafted features in time [14] or frequency domain [15] with shallow classifiers. However, there are several drawbacks of methods in this manner. Firstly, it is time consuming and laborious to extract the features. Secondly, it is difficult to design appropriate features oriented by different tasks. Thirdly, using features of pre-defined types can restrict capability of the representation to serve different applications and may result in capturing characteristics of minor importance for a task. Fourthly, as for vision and tactile sensing, acoustic features are present in a hierarchical structure, therefore, the use of handcrafted features and shallow classifiers will cause information

Shan Luo (corresponding author), Leqi Zhu, Kaspar Althoefer, Hongbin Liu are with the Centre for Robotics Research, Department of Informatics, King's College London, WC2R 2LS, UK (e-mail: shan.luo@kcl.ac.uk, leqi.zhu@kcl.ac.uk, kalthoefer@kcl.ac.uk, hongbin.liu@kcl.ac.uk).
loss. To learn abstract and hierarchical features automatically from raw sound data, in this paper we propose to apply deep learning for acoustic object recognition. The contributions of this paper can be summarized as:

1) A novel method to recognize objects is provided by using deep learning based on acoustic data.
2) The potential prospects of deep learning in acoustic based object recognition has been investigated and explored.
3) Different model parameters in deep architecture that would affect the recognition performance have been evaluated.

As depicted in Fig. 1, the acoustic data was generated by striking a marker pen on the test object. As a result, a vector of $1 \times N$ is acquired that is fed as the input to the deep network. The remainder of the paper is organized as follows. The related work on acoustic object recognition and deep learning based object recognition is reviewed in Section II. In Section III, the principle of Stacked Denoising Autoencoder (SDAE) and its application in audio based object recognition is introduced. And data collection procedure is presented in the following section. The results using deep learning are provided and the effect of different model parameters on the recognition performance is evaluated in Section V. It is compared with traditional sound recognition methods in Section IV. Finally, the paper is concluded in the last section.

II. RELATED WORK

A. Acoustic object recognition

In favor of object recognition based on visual and tactile information, less research has been done on acoustic-based object recognition. Nevertheless, audition is equally important as the sense of touch and vision, especially in the dark or hazardous environment. Burst et al. showed the feasibility to achieve object classification through sound information when struck the object [15]. Two most significant spikes in the power spectrum were extracted as features and their frequency coordinates were taken as inputs to a minimum-distance classifier. A reasonable classification accuracy of 94% was achieved, however, only five test objects were used. Followed this work, in [12] and[13], both the spectral content and decay rate were exploited to achieve the perception of object materials from contact sounds. In [16], the recognition of objects was based on the distributions of characteristic resonances and decay rates. In these works, actions of a single type, e.g., striking, were taken. Different from them, Sinapov et al. introduced a series of exploration behaviors, including shaking, grasping, dropping, tapping and pushing, to obtain different sounds from household objects [17]. Self-organizing Map (SOM) was implemented for feature extraction and k-Nearest Neighbor and SVM classifiers were used for classification. The work was studied further in [18] by integrating acoustic data with proprioceptive torque feedback and a better recognition rate was achieved. In a recent study [19], spectral energies were taken as the base features and a general Fourier domain analysis borrowed from the speech signal analysis literature was applied. In this work, the acoustic signals were generated by the interaction of a dexterous hexapod robot with the surfaces of different materials. In all the above works, complex preprocessing process had to be applied to eliminate the spurious signals and handcrafted features and shallow classifiers were utilized.

B. Deep learning based object recognition

As deep learning can extract higher-level representation of sensory inputs, it has attracted increasing attention in object recognition and shown promising results in different applications. To recognize the objects present in natural images, Krizhevsky et al. took the raw image pixels as inputs and trained a large, deep convolutional neural network to classify the objects in the ImageNet data set and enhanced the state-of-art recognition rate from 73.9% to 84.7% [20]. A more recent work was to employ deep convolutional neural networks to learn hierarchical features from RGB-D images for object recognition and pose estimation [21], and it also showed superior results than the traditional methods. In the view of tactile sensing, Schmitz et al. [22] applied deep learning in tactile object recognition and a dramatic performance improvement was observed in classifying 20 different objects compared to using traditional neural networks. However, to the best knowledge of the authors, there is still no work been done on acoustic object recognition by employing deep learning.

III. METHODOLOGY

The deep learning framework for the acoustic object recognition consists of two phases. The first phase is to train each deep network layer as a denoising autoencoder (DAE) by unsupervised pre-training in a layerwiser manner. The second phase is to stack the latent representations learned in the first phase to form a deep network that is fine-tuned as a whole using back propagations. In this phase, only the encoding part of each autoencoder in the first phase is considered. In the deep network the nodes in the input layer are the raw acoustic data and in the output layer are object classes. Based on the learned deep network, the test objects can be classified. Both phases are introduced in detail as follows.

A. Unsupervised pre-training

To begin with, the autoencoder, the base of DAE, is first introduced. It first maps the input vector $x$ into latent representation $y$ through transformation:

$$y = s(Wx + b),$$

where $s$ is the activation function (hyperbolic tangent $\tanh$ in our experiment), $W$ and $b$ are weight matrix and offset vector respectively. It is called encoding and the latent representation can be treated as a compressed representation of the input. After the encoding process, the latent representation is mapped back into a reconstruction $\hat{x}$ through a similar transformation:

$$\hat{x} = s(W'y + b').$$

It is called decoding and the output vector $\hat{x}$ can be interpreted as a prediction of the inputs $x$, given the latent representation. In this sense, the output layer has equally many nodes as the input layer. An autoencoder tends to minimize the error in
reconstructing its input $x$, i.e., to make the output $\hat{x}$ close or equal to input $x$. In the first autoencoder the input is the raw acoustic data and the obtained its latent representation is fed as input to the second autoencoder layer. In this manner, the latent representations of the other autoencoders are acquired.

The denoising autoencoder is a stochastic variant of the classical autoencoder. As in traditional autoencoders, it is still aimed to minimize the reconstruction loss between the input vector $x$ and its reconstruction from $y$. The difference is that $y$ is acquired from the transformation of a corrupted input, as shown in Fig. 2. It tries to undo the effect of a corruption process stochastically applied to the input of the autoencoder while preserving the information encoded in the input. In other words, a DAE is trained to reconstruct a “repaired” input from the corrupted input and make the latent representations become more robust features [23]. As can be seen in Fig. 2, this can be done by adding random noise into original input $x$, i.e., setting some of the inputs to zero.

Fig. 2. An illustration of the denoising autoencoder structure. The input $x$ is corrupted and the encoder is aimed to reconstruct $x$ from the corrupted input.

B. Supervised fine-tuning

Once all layers are pre-trained through DAEs, the deep network is constructed by stacking all the latent representation layers together as shown in Fig. 3. The input layer nodes are the raw acoustic data that are present in $1 \times N$ vectors, and the output nodes are object classes from 1 to $N_{obj}$ that is the number of objects. The entire network is then fine-tuned in a supervised manner to minimize the error in predicting the object labels using back propagation. More details can be found in [23].

IV. DATA COLLECTION

In our experiments, each data collection trial was carried out as follows. The test object was struck with a plastic marker pen and the generated impact sounds were recorded by Matlab in mono channel with the microphone of a laptop. For each trial, the object is struck at different places. The sampling frequency was 8 kHz and the recording time for each trial was two seconds. As a result, a series of 16000 data points in the range of [-1, 1] can be gained. To trim the redundant information in the data, only 500 points starting from the peak value of each sound signal, which can cover the whole knocking process for each trial, were taken as the input for deep learning model, therefore, $N=500$. The data collection process was conducted for 120 times for each object, with the first 100 times as the training phase and the remaining 20 times as the test phase. In total, thirty objects taken from the daily life were utilized in our experiments, as depicted in Fig. 4. It can be noticed that there are some objects of similar properties. For instance, object 1 and 2 are filled and unfilled bottles respectively; they have same surface properties but have different internal properties. It is very difficult to distinguish them by using visual appearance or static touch. For humans, it is very easy to

utilize the impact sound generated by striking to differ one from the other. Therefore, the robot is also expected to possess such capacity by employing our deep learning framework. For the purpose of minimizing the influence of noise, all experiments were performed in a relatively quiet room.

![Deep Structure Diagram](Fig. 3. An illustration of the deep structure. In this case, there are two hidden layers which are latent representations trained by denoising autoencoders separately. The input and output layers of the deep network are raw acoustic data and object classes respectively.)

V. EXPERIMENT RESULTS AND ANALYSIS

As presented in Section III, the SDAE model contains two main parts, the pre-training phase and the fine-tuning phase. To make the results more robust the classification process was conducted five times and the mean values of the results were taken. As a result of the structure of the collected data, in the deep network there were 500 nodes in the input layer and 30 nodes in the output layer. It was investigated how different parameters in deep learning model would affect the recognition performance that were number and layout of hidden layers, number of hidden nodes, number of iterations at pre-training and fine-tuning phases, learning rates, and the setting of with or without denoising. The parameters were optimized as shown in Table I according to the following three tips mentioned in [24]:

1) Adjust one single parameter at one time;
2) Scale consideration (e.g. learning rate of 0.1 and 0.2 may not differ much, but of 0.1 and 0.01 may have significant difference);
3) Computational considerations.

<table>
<thead>
<tr>
<th>Parameter</th>
<th>Value</th>
</tr>
</thead>
<tbody>
<tr>
<td>Number of hidden layers</td>
<td>3</td>
</tr>
<tr>
<td>Layout of hidden layers</td>
<td>Parallel</td>
</tr>
<tr>
<td>Number of hidden nodes</td>
<td>200</td>
</tr>
<tr>
<td>Unsupervised pre-training</td>
<td>500</td>
</tr>
<tr>
<td>Supervised fine-tuning</td>
<td>100</td>
</tr>
<tr>
<td>Learning rates</td>
<td>0.1</td>
</tr>
<tr>
<td>Denoising or not</td>
<td>Denoising</td>
</tr>
</tbody>
</table>

The effect of the number of hidden layers was investigated and interestingly it was found that more layers could out always yield superior recognition performance, as shown in Fig. 5. As the number of hidden layers was increased from 1 to 3, the recognition performance was enhanced. The probable reason for it is that more latent representations can extract more abstract features from the raw data at this stage. However, as the number of hidden layers was greater than 3, the recognition performance deteriorated, which could be interpreted as being affected by the excessive description. Therefore, a setting of three hidden layers
was selected in our study. The other parameters were as in Table I.

![Fig. 5. Recognition rates with various number of hidden layers.](image)

There are three types of layout of nodes in hidden layers: 1). increasing size, which is present in a shape of pyramid; 2). parallel size, in which all hidden layers have the same number of nodes; 3). decreasing size, which is present in a shape of inverted pyramid. The effect of these three layout types was investigated and the results are shown in Table II. It should be noted that the first layer and last layer in all cases are for the sensory input (500 nodes) and object classes (30 nodes) respectively. It can be observed that the parallel structure performed the best. It means that the parallel layout of hidden layers is more suitable for acoustic object recognition. Hence, parallel structure was also used in the subsequent tests.

**TABLE II RECOGNITION RATES WITH DIFFERENT LAYER STRUCTURE**

<table>
<thead>
<tr>
<th>Layout</th>
<th>Recognition rate</th>
</tr>
</thead>
<tbody>
<tr>
<td>500-100-200-300-30</td>
<td>4.50% (overfitting)</td>
</tr>
<tr>
<td>500-200-200-300-30</td>
<td>91.50%</td>
</tr>
<tr>
<td>500-300-200-100-30</td>
<td>73.83%</td>
</tr>
</tbody>
</table>

![Fig. 6. Recognition rates with various number of nodes in each hidden layer.](image)

Compared to the layer structure, it was found that the variance of the number of nodes used in each layer had less effect on the recognition performance. It can be seen in Fig. 6 that the recognition rate has a slight change when the number of nodes in each hidden layer is increased from 200 to 400. However, the computational expense would increase as more nodes present in each layer. Therefore, 200 nodes were present in each hidden layer.

The number of pre-training epochs was proved to be important for the recognition performance. As shown in Fig. 7, the recognition rates were increased as the number of pre-training epochs was incremented. It means that the more pre-training epochs are taken the better representations can be extracted from the raw data. But the performance levels off when the number of pre-training epochs arrived at 500. Hence, 500 epochs were taken for the pre-training phase.

![Fig. 7. Recognition rates with different pre-training epochs.](image)

Nevertheless, the epochs at the fine-tuning stage was found to have less effect on the recognition performance. As illustrated in Fig. 8, there was a small difference in the achieved recognition rates. Also taking the computational expense into consideration, 100 epochs were taken for the fine-tuning phase.

![Fig. 8. Recognition rates with different fine-tuning epochs.](image)

The learning rate of the pre-training phase was also considered as a significant parameter and its effect on the recognition performance is shown in Fig. 9. It could be divided into two phases: as the learning rate was incremented from 0.01 to 0.1 the
recognition rate was increased whereas when the learning rate was greater than 0.1 the recognition performance deteriorated dramatically. Hence, the learning rate was set as 0.1 in our experiment.

In addition, we investigated the impact of the process of denoising in the pre-training phase on the recognition performance. It was observed that the framework with denoising outperformed the one without denoising, with an improvement of 3.75% in the recognition rate. It indicates that the inclusion of denoising makes the learned deep learning model more robust.

Based on the above discussions, the optimized parameters were obtained as listed in Table I. As a result, an overall classification accuracy of 91.50% was achieved and a confusion matrix is shown in Fig. 10. It proves that our proposed deep learning framework can exploit the latent feature representations of the raw acoustic data and the objects can be recognized accurately. It can be observed that some objects that are difficult to distinguish by using vision or tactile sensing, e.g., filled and unfilled bottles (objects 1 and 2), can be classified successfully. On the other hand, only a few of the objects are assigned to wrong labels, e.g., some observations of the paper (object 20) are wrongly concluded to be from the kettle (object 5).

VI. COMPARISON WITH TRADITIONAL METHODS

As has been mentioned in Section I, traditional acoustic recognition is achieved by using handcrafted features with shallow classifiers. It is conducted in this section and compared with our proposed deep learning framework. In this paper, we used the Mel-Frequency Cepstral Coefficients (MFCCs) [25] and its first and second differential as features. It can well describe the nonlinear characteristics of human ear frequency and it is popular in traditional sound processing.

A Hanning window with a 32-ms fixed frame rate was first applied to the acquired acoustic signal to perform a Fourier transform. After that, we used 12th-order MFCC together with its first and second temporal derivatives as features. As a result, each feature was a 36 dimensional vector. The features were then implemented with a SVM classifier using LibSVM [26]. The best recognition rate using this method was only 58.22%. In addition, complex preprocessing process has to be applied and feature extraction process need to be designed elaborately.

Moreover, we utilized the raw acoustic data as the input of the SVM classifier. The recognition results using different methods are listed in Table III. It was surprising that a high recognition rate of 91.50% was achieved, much better than the case using MFCC features. A probable reason is that the original structure of the acoustic data appears to be more distinctive for shallow classifiers. But the recognition performance was still inferior to that of our proposed deep learning framework.

### TABLE III RECOGNITION RATES WITH DIFFERENT METHODS

<table>
<thead>
<tr>
<th>Method</th>
<th>Recognition rate</th>
</tr>
</thead>
<tbody>
<tr>
<td>Deep learning</td>
<td>91.50%</td>
</tr>
<tr>
<td>SVM with MFCC features</td>
<td>58.22%</td>
</tr>
<tr>
<td>SVM with raw data</td>
<td>82.00%</td>
</tr>
</tbody>
</table>

We also compared the time for classifying test objects that matters more when applied in real time applications. All algorithms were implemented in MATLAB and executed on a laptop with a 1.4Ghz Intel Core i5 processor and 4GB DDR3-1600 RAM. The time taken to classify test objects (excluding time for training the deep model) using our proposed deep

![Fig. 10. Confusion matrix of object recognition. The ground truths of the object labels are listed in the vertical axis while estimations are listed in the horizontal axis. The object labels are consistent with the ones in Fig. 4.](image-url)
learning framework was found to be much shorter. For classifying the thirty objects (20 trials for each), the minimum time taken was 0.03189s and the maximum time taken was 0.05450s using the deep learning framework. In comparison, by using SVM with raw data, the minimum time taken was 0.368843s. Hence, the classification using deep learning framework was at least 6.77 faster than the latter, as shown in Table IV. This inspiring result indicates that when a larger dataset is investigated, the strength of deep learning will be more revealed.

### Table IV Time taken for classifying test objects

<table>
<thead>
<tr>
<th>Method</th>
<th>Time/s</th>
</tr>
</thead>
<tbody>
<tr>
<td>Deep learning</td>
<td>0.05450 (longest)</td>
</tr>
<tr>
<td>SVM with raw data</td>
<td>0.5353 (shortest)</td>
</tr>
</tbody>
</table>

VII. CONCLUSIONS AND FUTURE WORK

This paper proposes a deep learning based method for the acoustic object recognition. Based on Stacked Denoising Autoencoders through both unsupervised pre-training and supervised fine-tuning, a multi-layer nonlinear mapping structure of deep network is trained to automatically extract high-level and more abstract features from the original acoustic data. It is proved that this deep learning based method can achieve better recognition performance compared to the traditional method using handcrafted features with a shallow classifier. It can be seen that the recognition rate increased 33.28% without complex feature extraction process through deep learning. It is also worth to note that the test time is dramatically faster using deep learning than using traditional method. In addition, various parameters in the deep learning network were investigated. In our experiment, there was no clear evidence to show that more layers would lead to better recognition performance. The parameters, including layout of hidden layers, number of hidden nodes, number of iterations at pre-training and fine-tuning phases, learning rates, and the setting of with or without denoising, were also studied.

There are several branches for future research. Compared to the large image recognition dataset like ImageNet, our dataset might not be big enough, thus it is planned to increase our dataset in the future work, not only by increasing the number of trials of striking, but also by enlarging the number of test objects. Deeper neural networks have achieved good results in the image classification task, e.g., the deep residue nets achieved 3.57% error on the ImageNet test set. Thus, it is also worth trying deeper neural nets in the acoustic object recognition [27]. In the current work most of the objects we choose are with relatively solid surface. However, for the soft or deformable objects such as clothes and teddy bear, it might become challenging because striking soft objects can only generate tiny impact sound. Hence, another extension of our work could be combining the auditory cues with other sensing information such as texture through tactile sensing.

REFERENCES

Shan Luo is currently a Research Fellow at University of Leeds and a Visiting Scientist at Computer Science and Artificial Intelligence Laboratory (CSAIL), Massachusetts Institute of Technology (MIT). He received the B.Eng. degree in Automatic Control from China University of Petroleum, Qingdao, China, in 2012. He was awarded the Ph.D. degree in Robotics from Kings College London, UK, in 2016. His research interests include tactile sensing, object recognition and computer vision.

Hongbin Liu (M-07) is currently a Lecturer (Assistant Professor) in the Centre for Robotics Research, Department of Informatics, Kings College London, UK, where he is leading the Compliant Robotics and Machine Haptics Lab. He received BEng in 2005 from the Northwestern Polytechnic University, China, and MSc in Mechatronics, PhD in robotics in 2006 and 2010 both from Kings College London, UK. He is a member of IEEE and Technical Committee Member for IEEE.
EMBS BioRobotics. His research is focusing on creating the robot’s haptic perception and developing for compliant and flexible robots for efficient physical interaction with the environment.