Interactive Lungs Auscultation with Reinforcement Learning Agent

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Abstract: To perform a precise auscultation for the purposes of examination of respiratory system normally requires the presence of an experienced doctor. With most recent advances in machine learning and artificial intelligence, automatic detection of pathological breath phenomena in sounds recorded with stethoscope becomes a reality.

But to perform a full auscultation in home environment by layman is another matter, especially if the patient is a child. In this paper we propose a unique application of Reinforcement Learning for training an agent that interactively guides the end user throughout the auscultation procedure. We show that intelligent selection of auscultation points by the agent reduces time of the examination fourfold without significant decrease in diagnosis accuracy compared to exhaustive auscultation.

1 INTRODUCTION

Lung sounds auscultation is the first and most common examination carried out by every general practitioner or family doctor. It is fast, easy and well known procedure, popularized by Laennec (Hyacinthe, 1819), who invented the stethoscope. Nowadays, different variants of such tool can be found on the market, both analog and electronic, but regardless of the type of stethoscope, this process still is highly subjective. Indeed, an auscultation normally involves the usage of a stethoscope by a physician, thus relying on the examiner’s own hearing, experience and ability to interpret psychoacoustical features. Another strong limitation of standard auscultation can be found in the stethoscope itself, since its frequency response tends to attenuate frequency components of the lung sound signal above nearly 120 Hz, leaving lower frequency bands to be analyzed and to which the human ear is not really sensitive (Sovijrvi et al., 2000) (Sarkar et al., 2015). A way to overcome this limitation and inherent subjectivity of the diagnosis of diseases and lung disorders is by digital recording and subsequent computerized analysis (Palaniappan et al., 2013).

Historically many efforts have been reported in literature to automatically detect lung sound pathologies by means of digital signal processing and simple time-frequency analysis (Palaniappan et al., 2013). In recent years, however, machine learning techniques have gained popularity in this field because of their potential to find significant diagnostic information relying on statistical distribution of data itself (Kandaswamy et al., 2004). Palaniappan et al. (2013) report state of the art results are obtained by using supervised learning algorithms such as support vector machine (SVM), decision trees and artificial neural networks (ANNs) trained with expert-engineered features extracted from audio signals. However, more recent studies (Kilic et al., 2017) have proved that such benchmark results can be obtained through end-to-end learning, by means of deep neural networks (DNNs), a type of machine learning algorithm that attempts to model high-level abstractions in complex data, composing its processing from multiple non-linear transformations, thus incorporating the feature extraction itself in the training process. Among the most successful deep neural network architectures, convolutional neural networks (CNNs) together with recurrent neural networks (RNNs) have been shown to be able to find useful features in the lung sound signals as well as to track temporal correlations between repeating patterns (Kilic et al., 2017).
However, information fusion between different auscultation points (APs) and integration of decision making processes to guide the examiner throughout the auscultation seems to be absent in the literature. Reinforcement Learning (RL), a branch of machine learning inspired by behavioral psychology (Sleight and Loewenstein, 2014), can possibly provide a way to integrate auscultation path information, interactively, at data acquisition stage. In the common RL problem setting, the algorithm, also referred as agent, learns to solve complex problems by interacting with an environment, which in turn provides positive or negative rewards depending on the results of the actions taken. The objective of the agent is thus to find the best policy, which is the best action to take, given a state, in order to maximize received reward and minimize received penalty. We believe that RL framework is the best choice for the solution of our problem, i.e finding the lowest number of APs while maintaining a minimum acceptable diagnosis accuracy. As a result, the agent will learn what are the optimal locations to auscultate the patient and in which order to examine them. As far as we know, this is the first attempt to use RL to perform interactive lung auscultation.

RL has been successful in solving a variety of complex tasks, such as computer vision (Bernstein et al., 2018), video games (Mnih et al., 2013), speech recognition (Kato and Shinozaki, 2017) and many others. RL can also be effective in feature selection, defined as the problem of identifying the smallest subset of highly predictive features out of a possibly large set of candidate features (Fard et al., 2013). A similar problem was further investigated (Bi et al., 2014) where the authors develop a Markov decision process (MDP) (Puterman, 1994) that through dynamic programming (DP) finds the optimal feature selecting sequence for a general classification task. Their work motivated us to take advantage of this framework with the aim of applying it to find the lowest number of APs, i.e the smallest set of features, while maximizing the accuracy in classifying seriousness of breath phenomena detected during auscultation, which is in turn directly proportional to diagnosis accuracy.

This work is organized as follows. Section 2 recalls the mathematical background useful to follow the work. Section 3 formally defines our proposed solution and gives a systematic overview of the interactive auscultation application. Section 4 describes the experimental framework used to design the interactive agent and evaluate its performance. Section 5 shows the results of our experiments, where we compare the interactive agent against its static counterpart, i.e an agent that always takes advantage of all auscultation points. Finally, Section 6 presents our conclusions.

2 MATHEMATICAL BACKGROUND

2.1 Reinforcement Learning

The RL problem, originally formulated in (Sutton, 1988), relies on the theoretical framework of MDP, which consists on a tuple of \((S, A, P_s, \gamma, R)\) that satisfies the Markov property (Puterman, 1994). \(S\) is a set of environment states, \(A\) a set of actions, \(P_s\) the state (given an action) transitions probability matrix, \(\gamma\) the discount factor, \(R(s)\) the reward (or reinforcement) of being in state \(s\). We define the policy \(\pi(s)\) as the function, either deterministic or stochastic, which dictates what action to take given a particular state. We also define a value function that determines the value of being in a state \(s\) and following the policy \(\pi\) till the end of one iteration, i.e an episode. This can be expressed by the expected sum of discounted rewards, as follows:

\[
V^\pi(s) = E[R(s_0) + \gamma R(s_1) + \gamma^2 R(s_2) + \ldots | s_0 = s, \pi] 
\]

(1)

where \(s_0, s_1, s_2, \ldots\) is a sequence of states within the episode. The discount factor \(\gamma\) is necessary to moderate the effect of observing the next state. When \(\gamma\) is close to 0, there is a shortsighted condition; when it tends to 1 it exhibits farsighted behaviour (Sutton and Barto, 1998). For finite MDPs, policies can be partially ordered, i.e \(\pi \preceq \pi'\) if and only if \(V^\pi(s) \geq V^\pi'(s)\) for all \(s \in S\). There is always at least one policy that is better than or equal to all the others. This is called optimal policy and it is denoted by \(\pi^*\). The optimal policy leads to the optimal value function:

\[
V^*(s) = \max_\pi V^\pi(s) = V^{\pi^*}(s) 
\]

(2)

In the literature the algorithm solving the RL problem (i.e finding \(\pi^*\)) is normally referred as the agent, while the set of actions and states are abstracted as belonging to the environment, which interacts with the agent signaling positive or negative rewards (Figure 1). Popular algorithms for its resolution in the case of finite state-space are Value iteration and Policy iteration (Sutton, 1988).

![Figure 1: RL general workflow: the agent is at state \(s_0\), with reward \(R[s_0]\). It performs an action \(a\) and goes from state \(s_0\) to \(s_1\) getting the new reward \(R[s_1]\).](image-url)
2.2 Q-Learning

Q-learning (Watskin and Dayan, 1992) is a popular algorithm used to solve the RL problem. In Q-learning, actions \( a \in A \) are obtained from every state \( s \in S \) based on an action-value function called \( Q \) function, \( Q : S \times A \rightarrow \mathbb{R} \), which evaluates the quality of the pair \((s, a)\).

The Q-learning algorithm starts arbitrarily initializing \( Q(s, a) \); then, for each episode, the initial state \( s \) is randomly chosen in \( S \), and \( a \) is taken using the policy derived from \( Q \). After observing \( r \), the agent goes from state \( s \) to \( s' \) and the \( Q \) function is updated following the Bellman equation (Sammut and Webb, 2010):

\[
Q(s, a) \leftarrow Q(s, a) + \alpha[r + \gamma \max_{a'} Q(s', a') - Q(s, a)]
\]

(3)

where \( \alpha \) is the learning rate that controls algorithm convergence and \( \gamma \) is the discount factor. The algorithm proceeds until the episode ends, i.e. a terminal state is reached. Convergence is reached by recursively updating values of \( Q \) via temporal difference incremental learning (Sutton, 1988).

2.3 Deep Q network

If the states are discrete, \( Q \) function is represented as a table. However, when the number of states is too large, or the state space is continuous, this formulation becomes unfeasible. In such cases, the \( Q \)-function is computed as a parameterized non-linear function of both states and actions \( Q(s, a; \theta) \) and the solution relies on finding the best parameters \( \theta \). This can be learned by representing the Q-function using a DNN as shown in (Mnih et al., 2013) (Mnih et al., 2015), introducing deep Q-networks (DQN).

The objective of a DQN is to minimize the mean square error (MSE) of the Q-values:

\[
L(\theta) = \frac{1}{2} [r + \max_{a'} Q(s', a'; \theta) - Q(s, a; \theta)]^2
\]

(4)

\[
J(\theta) = \max_\theta [L(\theta)]
\]

(5)

Since this objective function is differentiable w.r.t \( \theta \), the optimization problem can be solved using gradient based methods, e.g. Stochastic Gradient Descent (SGD) (Bottou et al., 2018).
Finally, reinforcement signals ($R$) are designed in the following way: rewards are given when the predicted diagnosis status is correct, penalties in the opposite case. Moreover, in order to discourage the agent of using too many points, a small penalty is provided for each additional auscultated point. The best policy for our problem is thus embodied in the best auscultation path, encoded as sequence of most informative APs to be analyzed, which should be as shortest as possible.

3.3 Application

The interactive auscultation application consists of two entities: the pair digital stethoscope and smartphone, used as the interface for the user to access the service; and a remote server, where the majority of the computation is done and where the agent itself resides. An abstraction of the entire system is depicted in Figure 2.

The first element in the pipeline is our proprietary stethoscope (StethoMe, 2018). It is a digital and wireless stethoscope similar to Littmann digital stethoscope (Littmann 3200, 2009) in functionality, but equipped with more microphones that sample the signal at higher sampling rate, which enables it to gather even more information about the patient and background noise. The user interacts with the stethoscope through a mobile app installed on the smartphone, connected to the device via Bluetooth Low Energy protocol (Gomez et al., 2012). Once the auscultation has started, a high quality recording of the auscultated point is stored on the phone and sent to the remote server. Here, the signal is processed and translated into a fixed number of features that will be used as input for the agent. The agent predicts which is the best action to perform next: it can be either to auscultate another point or, if the confidence level is high enough, return predicted patient’s status and end the examination. Agent’s decision is being made dynamically after each recording, based on breath phenomena detected so far. This allows the agent to make best decision given limited information which is crucial when the patient is an infant and auscultation gets increasingly difficult over time.

4 EVALUATION

This section describes the experimental framework used to simulate the interactive auscultation application described in Subsection 3.3. In particular, in Subsection 4.1 the dataset used for the experiments is described. In Subsection 4.2 a detailed description of the feature extraction module already introduced in Subsection 3.2 is provided. In Subsection 4.3 the interactive agent itself is explained, while in Subsection 4.4 the final experimental setup is presented.

4.1 Dataset

Our dataset consists of a total of 570 real examinations conducted in the Department of Pediatric Pulmonology of Karol Jonscher Clinical Hospital in Poznań (Poland) by collaborating doctors. Data col-
Figure 3: Feature extraction module: audio signal is first converted to spectrogram and subsequently fed to a CRNN, which outputs a prediction raster of 5 classes: inspiration, expiration, wheezes, crackles and noise. This raster is then post-processed with the objective of extracting values representative of detection and intensity level of the critical phenomena, i.e wheezes and crackles. More specifically, maximum probability value and relative duration of tags are computed per each inspiration/expiration and the final features are computed as the average of these two statistics along all inspirations/expiration.

4.2 Feature Extractor

The features for the agent are extracted by a feature extractor module that is composed of three main stages, schematically depicted in Figure 3 at the beginning of the pipeline, the audio wave is converted to its magnitude spectrogram, a representation of the signal that can be generated by applying short time fourier transform (STFT) to the signal (a). The time-frequency representation of the data is fed to a convolutional recurrent neural network (b) that predicts breath phenomena events probabilities in form of predictions raster. Raster is finally post-processed in order to extract 8 interesting features (c).

4.2.1 CRNN

This neural network is a modified implementation of the one proposed by Çakir et al. (2017), i.e a CRNN designed for polyphonic sound event detection (SED). In this structure originally proposed by the convolutional layers act as pattern extractors, the recurrent layers integrate the extracted patterns over time thus providing the context information, and finally the feedforward layer produce the activity probabilities for each class (Çakir et al., 2017). We decided to extend this implementation including dynamic routing (Sabour et al., 2017) and applying some key ideas of Capsule Networks (CapsNet), as suggested in recent advanced studies (Vesperini et al., 2018) (Liu et al., 2018). The CRNN is trained to detect 5 types of sound events, namely: inspirations, expirations, wheezes, crackles, and noise.

Table 1: Number of examinations for each of the classes.

| Label | Description             | N_examinations |
|-------|-------------------------|----------------|
| 0     | no auscultatory changes - no alarm | 200            |
| 1     | innocent auscultatory changes - no alarm | 85             |
| 2     | significant auscultatory changes - alarm | 285            |
Figure 4: Interactive agent learning curves: in the very first episodes the agent randomly guesses the state of the patient, without auscultating any point. Next comes the exploration phase when the agent auscultates many points, often reaching the 12-point limit which results in high penalties. Finally, as the agent plays more and more episodes, it starts learning the optimal policy using fewer and fewer points until he finds the optimal solution.

4.2.2 Raster Post-processing

Wheezes and crackles are the two main classes of pathological lung sounds. The purpose of raster post-processing is to extract a compact representation that will be a good description of their presence/absence and level of intensity. Thus, for each inspiration and expiration event we calculate two values for each pathological phenomena (wheeze and crackle): maximum probability within said inspiration/expiration and relative duration after thresholding (the level in which the inspiration/expiration is filled, or covered with the pathological phenomena). All extracted values are then averaged across all inspirations and expirations separately. We therefore obtain 8 features: average maximum wheeze probability on inspirations (1) and expirations (2), average relative wheeze length in inspirations (3) and expirations (4) and the same four features (5, 6, 7 and 8) for crackles.

4.3 Reinforcement Learning Agent

Our agent consists of a deep fully connected neural network. The network takes as input a state matrix of size 12 rows × 9 columns; then processes the input through 3 hidden layers with 256 units each followed by ReLU nonlinearity (Nair and Hinton, 2010); the output layer is composed by 15 neurons which represent expected rewards for each of the possible future actions. This can be either to request one of the 12 points to be auscultated, or declare one of the three alarm status, i.e predict one of the three labels and finish the examination. The state matrix is initially set to all zeros, and the $i_{th}$ row is updated each time $i_{th}$ AP is auscultated.

First 8 columns of state matrix correspond to eight features described in previous section, while the last value is a counter for the number of times this auscultation point was auscultated. At every interaction with the environment, the next action $a$ to be taken is defined as $\arg\max$ of the output vector. At the beginning of the agent’s training we ignore agent’s preferences and perform random actions, as the training proceeds we start to use agent’s recommended actions more and more often. For a fully trained model, we always follow agent’s instructions. The agent is trained to predict three classes, but classes 0 and 1, treated as agglomerated not alarm class, are eventually merged at evaluation phase.

The agent is trained with objective of minimizing Eq. 4 and Q-values are recursively updated by temporal difference incremental learning. There are two ways to terminate the episode: either make a classification decision, getting the reward/penalty that follows table 2; or reaching a limit of 12 actions which results in a huge penalty of $r = -10.0$. Moreover, when playing the game a small penalty of $r = -0.01$ is given for each requested auscultation, this is to encourage the agent to start the examination if it doesn’t expect any more information coming from continued auscultation.

Table 2: Reward matrix for Reinforcement Learning Agent final decisions.

| actual | predicted | 0  | 1 | 2 |
|--------|-----------|----|---|---|
| 0      | 2.0       | 0.0| -1.0|
| 1      | 0.0       | 2.0| -0.5|
| 2      | -1.0      | -0.5| 2.0|
4.4 Experimental Setup

We compared the performance of reinforcement learning agent, from now on referred to as interactive agent, to its static counterpart, i.e. an agent that always performs an exhaustive auscultation (uses all 12 APs). In order to compare the two agents we performed 5-fold cross validation for 30 different random splits of the dataset into training (365 auscultations), validation (91) and test (114) set. We trained the agent for 200 episodes, setting $\gamma = 0.93$ and using Adam optimization algorithm (Kingma and Ba, 2014) to solve Eq. \ref{eq:4} with learning rate initially set to 0.0001.

Both in validation and test phase, 0 and 1 labels were merged as single not alarm classes. Therefore results shown in the following refer to the binary problem of alarm detection: we chose as comparative metrics balanced accuracy (BAC) defined as un-weighted mean of sensitivity and specificity; and F1-score, harmonic mean of precision and recall, computed for each of the two classes.

5 RESULTS

In Table \ref{table:3} we show the results of the experiments we conducted. The interactive agent performs the auscultation using on average only 3 APs, effectively reducing the time of the examination 4 times. This is a very significant improvement and it comes at a relatively small cost of 2.5 percent point drop in classification accuracy.

Table 3: Results of experiments.

| Agent   | BAC   | $F_1_{\text{alarm}}$ | $F_1_{\text{not alarm}}$ | APs |
|---------|-------|----------------------|---------------------------|-----|
| Static  | 84.8% | 82.6%                | 85.1%                     | 12  |
| Interactive | 82.3% | 81.8%                | 82.6%                     | 3.2 |

Figure 4 shows learning curves of rewards and number of points auscultated by the agent. In the very first episodes the agent directly guesses the state of the patient, declaring the alarm value without auscultating any point. As soon as it starts exploring other possible action-state scenarios, it often reached the predefined limit of 12 auscultation points which significantly reduces its average received reward. However, as it plays more episodes, it starts converging to the optimal policy, using less points on average.

In order to assess the knowledge learned by the agent, we conducted a survey involving a total of 391 international experts. The survey was distributed among the academic medical community and in hospitals. In the survey we asked each participant to respond a number of questions regarding education, specialization started or held, assessment of their own skills in adult and child auscultation, etc. In particular, we asked them which points among the 12 proposed would be auscultated more often during an examination. Results of the survey are visible in Figure 5 where we compare collected answers with most used APs by the interactive agent. It’s clear that the agent was able to identify which APs carry the most information and are the most representative to the overall patient’s health status. This is the knowledge that all human experts gain from many years of clinical practice. In particular the agent identified points 11 and 12 as very important. This finding is confirmed by the doctors who strongly agree that these are the two most important APs on patient’s back. On the chest both doctors and the agent often auscultate point number 4, but the agent prefers point number 2 instead of 3, probably due to the distance from the heart which is a major source of interference in audio signal.
The agent seems to follow two general rules during the auscultation: firstly, it auscultates points belonging both to the chest and to the back; secondly, it tries to cover as much area as possible, visiting not-subsequent points. For instance, the top 5 auscultation paths among the most repeating sequences that we observed are: [4, 9, 11], [8, 2, 9], [2, 11, 12], [7, 2, 8], [4, 11, 12]. These paths cover only 3% of the possible paths followed by the agent: this means the agent does not follow a single optimal path or even couple of paths, but instead uses a wide variety of paths depending on breath phenomena detected during the examination.

6 CONCLUSIONS

We have presented a unique application of reinforcement learning for lung sounds auscultation, with the objective of designing an agent being able to perform the procedure interactively in the shortest time possible.

Our interactive agent is able to perform an intelligent selection of auscultation points. It performs the auscultation using only 3 points out of a total of 12, reducing fourfold the examination time. In addition to this, no significant decrease in diagnosis accuracy is observed, since the interactive agent gets only 2.5 percent points lower accuracy than its static counterpart that performs an exhaustive auscultation using all available points.

Considering the research we have conducted, we believe that further improvements can be done in the solution proposed. In the near future, we would like to extend this work to show that the interactive solution can completely outperform any static approach to the problem. We believe that this can be achieved by increasing the size of the dataset or by more advanced algorithmic solutions, whose investigation and implementation was out of the scope of this publication.

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