Sleep Snoring Sound Recognition Based on Wavelet Packet Transform

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Snoring is a typical and intuitive symptom of the obstructive sleep apnea hypopnea syndrome (OSAHS), which is a kind of sleep-related disorder having adverse effects on people’s lives. Detecting snoring sounds from the whole night recorded sounds is the first but the most important step for the snoring analysis of OSAHS. An automatic snoring detection system based on the wavelet packet transform (WPT) with an eXtreme Gradient Boosting (XGBoost) classifier is proposed in the paper, which recognizes snoring sounds from the enhanced episodes by the generalization subspace noise reduction algorithm. The feature selection technology based on correlation analysis is applied to select the most discriminative WPT features. The selected features yield a high sensitivity of 97.27% and a precision of 96.48% on the test set. The recognition performance demonstrates that WPT is effective in the analysis of snoring and non-snoring sounds, and the difference is exhibited much more comprehensively by sub-bands with smaller frequency ranges. The distribution of snoring sound is mainly on the middle and low frequency parts, there is also evident difference between snoring and non-snoring sounds on the high frequency part.

Keywords: snoring recognition; wavelet packet transform; feature selection; machine learning.

1. Introduction

The obstructive sleep apnea hypopnea syndrome (OSAHS) is a chronic sleep-related disease affecting the general adult population ranging from 6% to 17% (Senaratna et al., 2017), which is characterized by intermittently partial or complete collapse of the upper airway, resulting in frequently sleep-disordered breathing events (SDB). This kind of disease greatly affects the quality of life and even is an independent risk factor for diseases such as neurocognitive dysfunction, arterial hypertension, metabolic disorders, and cerebrovascular disease (Wang et al., 2017; Hui et al., 2015; Dafna et al., 2013). The traditional and golden standard for clinically diagnosing OSAHS is Polysomnography (PSG) (Jiang et al., 2020) with multiple sensors that must be directly connected to the body to monitor serious biological signals during sleep. However, the complex equipment, professional technicist, time-consuming process, and expensive cost limiting its wide application, makes OSA a significant but underestimated threat to public health (Ayas, 2013). An inexpensive and reliable technology to diagnose OSAHS is urgently needed. Studies have indicated that snoring is a typical and intuitive symptom of OSAHS (Dafna et al., 2013; Hui et al., 2015; Jiang et al., 2020; Ng et al., 2008; Senaratna et al., 2017; Wang et al., 2017) reported in more than 80% of OSAHS patients (Kapur et al., 2002; Young et al., 1997), which has been reported to be a potential method to monitor OSAHS. It is a kind of sleep-related noise caused by oscillations of the soft tissue structures in the upper airways (Lechner et al., 2019) because of a reduction of the muscle tone and slackening of soft tissue narrowing down the upper airways.

Automatic extracting snoring episodes from recorded sleep sounds throughout the night, including breathing, speaking, and other noises, is the first but
the most important step during the whole process of analyzing snoring sounds, which has been studied by many studies (DAFNA et al., 2013; HWANG et al., 2015; LIM et al., 2019; NG et al., 2008; NONAKA et al., 2016; WANG et al., 2017). Most of their studies focused on differencing snoring sounds and non-snoring sounds from acoustic features derived from the time domain, frequency domain, and time-frequency domain. Specifically, JIANG et al. (2020) designed a snoring sound detection system based on a non-contact microphone that extracted 127-dimensional time and spectral features to obtain an accuracy of 98.2% in the validation group. The work of NG et al. (2008) achieved a high accuracy in the classification of snoring and non-snoring sounds using formant frequencies. CAVUSOGLU et al. (2007) implemented the snoring and non-snoring classification by the sub-band energy of sound episodes and robust linear regression with an accuracy of 90.2% from the combined dataset of 18 simple snorers and 12 OSAHS patients. NONAKA et al. (2016) developed the human auditory image model to extract snoring sounds automatically. These studies mainly focused on the low frequency part or some specific frequency bands rather than analysis of all sub-band, which might ignore the information on the high frequency part.

The wavelet transform (WT) is another way to divide the signal into sub-bands with different frequency ranges, which has been demonstrated to be effective in the speech signal and electroencephalogram processing (LI, ZHOU, 2016; WU et al., 2008; WANG et al., 2020). WANG et al. (2020) proposed a novel method of speaker-independent emotion recognition based on the wavelet packet analysis, which performed better than frequency features. LI and ZHOU (2016) implemented the classification of electrocardiograms using the wavelet packet entropy and random forests. WU et al. (2008) extracted features of electroencephalogram signals such as the energy of special sub-bands and corresponding coefficients of the wavelet packet decomposition, which had maximal separability according to the Fisher distance criterion. QIAN et al. (2016; 2017) adopted energy features derived from the wavelet packet transform (WPT) to discriminate snoring sounds from different snoring sites with much better performance than features derived from time and frequency domains. These works indicate that wavelet transform works effectively in the analysis of biological signals.

To explore the relationship between snoring and non-snoring sounds on different frequency bands, an automatic snoring detection system based on WPT features with an XGBoost (CHEN, GUESTRI, 2016) classifier was proposed in this study. The system includes three major steps. Firstly, the recorded sleep-related sounds were enhanced and segmented by a generalization subspace noise reduction algorithm, and signal presence probability based on energy, respectively. Then, WPT features from different wavelet functions and decomposition layers were extracted from segmented sound episodes and selected based on a series of correlation analyses. Finally, snoring sounds were detected from the trained XGBoost classifier. The contribution of the work incorporates: 1) it used WPT to extract sub-band features and yielded comparable accuracy in recognizing snoring sounds compared with existing related studies (ADESUUYI et al., 2022; ARSENALI et al., 2018; JIANG et al., 2020; SUN et al., 2022); 2) it discovered that signal would be exhibited much more comprehensively by sub-bands with smaller frequency ranges. And the difference between snoring and non-snoring sounds is getting more evident with the frequency range getting smaller, which is more beneficial for classifying; 3) it demonstrated that although the distribution of snoring sound is mainly on the low frequency part, the information on the high frequency part also cannot be ignored, where there is also evident difference between snoring and non-snoring sounds.

2. Material and methods

2.1. Data acquired

In this study, 24 subjects composed of simple snorers and OSAHS patients were selected from the First Affiliated Hospital of Guangzhou Medical University. All subjects have been informed and agreed with the monitoring process during the whole night. The detailed information about these subjects was described in Table 1. During sleeping, a microphone (RODE, NTG-3, Sydney, Australia) and a digital audio recorder (Rowland, Edirol R-44, Japan) were placed approximately 45 cm above the patient’s mouth and nose to record the original sleep sound signals for approximately seven hours, with a sampling rate of 44.1 kHz and 16-bit resolution. PSG equipment (Alice-5, Pittsburgh, Pennsylvania, USA) was simultaneously used to monitor the subject’s PSG signals.

| Table 1. Information (gender, age, apnea/hypopnea (AHI), and body mass index (BMI)) of the subject. |
|---|---|---|---|---|
| **Gender (M/F)** | Simple | Mild | Moderate | Severe |
| **Age (years)** | 25 ± 5 | 35 ± 5.05 | 46.6 ± 11.58 | 49.9 ± 9.36 |
| **AHI** | 3.8 ± 0.73 | 10.8 ± 4.13 | 21.69 ± 3.46 | 36.77 ± 3.83 |
| **BMI** | 27.5 ± 4.57 | 30.24 ± 0.87 | 35.09 ± 1.05 | 39 ± 1.48 |
2.2. Pre-processing

During the process of recording sleep-related sounds, the noncontact nature of signal acquisition is often susceptible to external noise distortion. The additive background noise is inevitably superposed to a snoring sound, which will affect the fidelity of signals. Most studies (DAFNA et al., 2013; JIANG et al., 2020; LIM et al., 2019; WANG et al., 2017) conducted the noise reduction process before its further analysis effectively. In the work of KARUNAJEEWA et al. (2008), different enhancement algorithms were implemented to yield different snoring recognition results. Different from common noise suppression methods, generalized subspace snoring signal enhancement based on the noise covariance matrix estimation was implemented (DING et al., 2021). It was verified by our previous work that this algorithm could well update noise in real-time by recursive averaging its past values adjusted by a time-varying smoothing parameter controlled by the snoring signal presence probability during the noise suppression process. Objective quality measurements and the spectrum analysis demonstrated that this method could reduce most background noise with less signal distortion. Moreover, the enhanced snoring signal was detected and segmented by the signal presence probability, which is determined by the ratio between the local energy of the noisy signal and its minimum within a specified time window to detect the sound episode.

Figure 1 shows the process of pre-processing including noise reduction and episode segmentation. The segmented episodes were further labeled as snoring sounds and non-snoring sounds based on PSG signals by ear-nose-throat (ENT) experts. 26561 labeled sound episodes including 17704 snoring sounds and 8857 non-snoring sounds were obtained from all 24 subjects. All labeled sounds were randomly divided into the training and test sets with proportions of 70% and 30%, respectively.

2.3. Feature extraction

The wavelet packet decomposition was applied to divide the signal into sub-bands with different frequency bands. Acoustic features including wavelet packet coefficients, log energy, Shannon entropy, wavelet transform cepstral coefficient, and sound pressure level based on sub-band signals were extracted for further analysis. All signals were framed by the hamming window function with 20 ms frame length and 50% overlap. Amplitude normalization was conducted to eliminate the influence of sound intensities. The statistic functions including the mean and variance of all frames in each signal were calculated to represent each signal. Table 2 shows the detailed information on the features.

<table>
<thead>
<tr>
<th>Feature</th>
<th>Description</th>
<th>Dimension (layer 4/layer 5/bark sub-band)</th>
</tr>
</thead>
<tbody>
<tr>
<td>Coefficient</td>
<td>Mean, variance value</td>
<td>32/64/34</td>
</tr>
<tr>
<td>Log-energy</td>
<td>Mean, variance value</td>
<td>32/64/34</td>
</tr>
<tr>
<td>Shannon entropy</td>
<td>Mean, variance value</td>
<td>32/64/34</td>
</tr>
<tr>
<td>Sound pressure level</td>
<td>Mean, variance value</td>
<td>32/64/34</td>
</tr>
<tr>
<td>Wavelet transform cepstral coefficient</td>
<td>Mean, variance value</td>
<td>26/26/26</td>
</tr>
<tr>
<td>Mel-frequency cepstral coefficient</td>
<td>Mean, variance value</td>
<td>26/26/26</td>
</tr>
</tbody>
</table>

2.3.1. Wavelet packet model

The WT (SHARMA et al., 2020) is a typical method to transform the time-domain audio signal into a time-frequency domain consisting of the continuous wavelet transform and discrete wavelet transform. The WT of the signal $x$ at the time $y$ and scale $z$ is defined by the inner product with a wavelet function: $W_f(y, z) = \langle x, u_{y,z} \rangle = \frac{1}{\sqrt{y}} \int_{-\infty}^{\infty} f(t) u^\ast\left(\frac{t - z}{y}\right) \, dt$, (1)

where $u(t)^\ast$ represents the complex conjugate of the wavelet function $u(t)$. WPT applies the transform
step on all frequency bands. It is calculated through time-domain filtering with a sub-signal representation obtained from frequency components with each sub-band. Figure 2 shows an integrated wavelet packet tree of a signal. The original signal is first decomposed into two sub-bands in the first decomposition layer: the low frequency part-1 (1, 0) and the high frequency part-2 (1, 1). Then the low frequency part-1 and part-2 will be further decomposed with increasing decomposition layers to obtain sub-bands with much finer frequency bandwidth. The frequency bandwidth of the k-th sub-band in the j-th decomposition layer is \( \frac{f_s}{2^j} \) Hz; \( f_s \) is the sampling rate with a value of 44.1 kHz in this work. With the increase of the decomposition layer, the finer the frequency is decomposed with much more sub-bands. There are \( N_j = 2^j \) sub-bands in the j-th layer. There are 16 sub-bands in layer 4 with a bandwidth of 1378 Hz and 32 sub-bands in layer 5 with a bandwidth of 689 Hz. Moreover, we constructed bark sub-bands from decomposition layers 4 and 5 of WPT with a dimension of 17, which is based on auditory characteristics of humans. The detailed composition of bark sub-bands is shown in the last layer of Fig. 2.

![Decomposition process of WPT and the construction of wavelet bark.](image)

In the work of Monson et al. (2014) “high frequency” referred to a frequency above about 5 kHz which has traditionally been neglected in speech research. The “middle and low frequency” is defined below 5 kHz in this paper. As Fig. 2 shows, the bark sub-band is constructed by sub-bands in layer 4 and layer 5 which is considered a common division method for audio signal processing that accurately matches the human ear’s auditory perception characteristics (Kornienko, Machusky, 2018). The bark and layer 5 decomposition structure have the same sub-band distributions in the middle and low frequency part (0–5.5 kHz) composed by (5,0)–(5,7), while layer 5 has much finer sub-bands in the high frequency. Studies have indicated that the energy of snoring sounds is mainly concentrated below 2 kHz (Perez-Padilla et al., 1993). The frequency range of the first sub-band at layer 4 is 0–1378 Hz that most snoring information locate in this frequency band. To explore the influence of the frequency bandwidth of the sub-band on the classification result, and the difference between snoring and non-snoring sounds in the middle and low frequency part and high frequency part, acoustic features extracted from layer 4, layer 5, and bark sub-bands are discussed in the work.

### 2.3.2. Wavelet packet coefficients

The coefficients by WPT can reveal the local characteristics of signals. The mean values of the coefficients of the k-th sub-band in the j-th layer are described as:

\[
 w_{j,k} = \frac{\sum v_{j,k,n}}{N_k}, \quad n = 1, 2, \ldots, N_k, \quad k = 1, 2, \ldots, 2^j, \tag{2}
\]

where \( N_k \) is the number of the coefficient component in the k-th sub-band with the value of 882; \( v_{j,k,n} \) represents the n-th coefficient component of the k-th sub-band in the j-th decomposition layer. There are \( N_j = 2^j \) coefficients in the j-th decomposition, which are \( w_{j,1}, w_{j,2}, \ldots, w_{j,2^j} \).

#### 2.3.3. Log-energy

The log-energy of the k-th sub-band signal in the j-th level can be calculated by:

\[
 \log E_{j,k} = \sum_{n=1}^{N_k} v_{j,k,n}^2. \tag{3}
\]

#### 2.3.4. Shannon entropy

The probability of the n-th coefficient at its corresponding node can be calculated by:

\[
 p_{j,k,n} = \frac{E_{j,k}}{E_{j,k}} = \frac{v_{j,k,n}^2}{\sum_{n=1}^{N_k} v_{j,k,n}^2}, \tag{4}
\]

moreover, Shannon entropy (SE) is defined by the probability distribution of energy \( p_{j,k,n} \) as Eq. (4), which is a measure of uncertainty associated with random variables in information theory:

\[
 SE_{j,k} = -\sum_{n=1}^{N_k} p_{j,k,n} \cdot \log (p_{j,k,n}). \tag{5}
\]

#### 2.3.5. Wavelet transform cepstral coefficient

WPT can be treated as a filter to divide the frequency to some sub-bands with the equal bandwidth, just like Mel-filter. Then the cepstral coefficient of the signal after the WPT filter can be calculated using discrete cosine transform (DCT), which is called the wavelet transform cepstral coefficient (WTCC):

\[
 \text{WTCC}_m = \sqrt{\frac{2}{N_j} \sum_{j=0}^{N_j} \log (E_{j,k}) \cos \left( \frac{\pi m (2j - 1)}{2N_j} \right)}, \tag{6}
\]

where \( N_j \) is the number of sub-bands in the j-th layer; \( m \) indicates the m-th DCT spectral line, which was set as 13 in the work.
2.3.6. Mel-frequency cepstral coefficients

The study (Jiang et al., 2020) has indicated that there are obvious differences between snoring and non-snoring sounds via Mel-frequency cepstral coefficients (MFCC). The MFCC of the original signal with 13-dimension is also extracted for snoring sound recognition.

2.4. Wavelet function

As described in Eq. (1), the WPT is based on the wavelet function. Different wavelet functions may result in different WPT features. There are many wavelet function families such as BiorSplines, Coiflets, Daubechies, Symlets, and so on. The Daubechies wavelet family (DB1, DB2, DB3, DB4, DB5, DB6, DB7, DB8, DB9, DB10) has been widely used in the processing of speech and other biological signals. Also, the work of Qian et al. (2017) indicated that the Daubechies function (DB3, DB10) performed better on recognition of a snoring site. In this paper, we explore their performance in differentiating snoring and non-snoring sounds.

3. Classification and result

3.1. Classification model

In the study, XGBoost classifier was adopted in this study. XGBoost is an improved algorithm with good performance, and high efficiency based on the gradient boosting decision that can construct boosted trees efficiently and operate in parallel. The core of the algorithm is the optimization of the value of the objective function (Torlay et al., 2017). The parameter of the XGBoost classifier is essential for classification performance. Based on a 10-fold cross-validation of the training set, the optimal parameter was obtained. The number of base trees was set as 400, the max depth of trees was 6, and the learning rate was 0.3. Other parameters were set as the default value of XGBoost in Scikit-learn (Pedregosa et al., 2011).

3.2. Feature selection

Feature selection is a vitally important step during the classification task because it can reduce the redundancy of features to improve the robustness of the model and reduce the computation complexity. In this paper, feature selection based on the correlation analysis is conducted to select distinguishing features. Two Pearson correlation coefficients were calculated including correlation coefficients between features and their related labels with a value of P1 and correlation coefficients among features with a value of P2. Features with high correlation with labels and low correlation with other features were selected by thresholds a and b, respectively. There were two steps for feature selection. Firstly, features were reserved if P1 was higher than a. Then, the reserved features were dropped out if P2 was higher than b to obtain relatively independent features. To fully make use of the limited dataset, the 10-fold cross-validation was used in the training set to optimize the model and select features. The threshold a and b were obtained by experiment to set as 0.8 and 0.7, respectively. Moreover, the effect of the decomposition levels and wavelet functions on the classification of snoring and non-snoring sound is explored.

3.3. Model evaluation

To evaluate the performance of the proposed recognition system of snoring sound, evaluating indexes such as sensitivity, accuracy, precision, and F1 score are expressed as follows:

\[
\text{Accuracy} = \frac{\text{TP} + \text{TN}}{\text{TP} + \text{FP} + \text{TN} + \text{FN}},
\]

\[
\text{Sensitivity} = \frac{\text{TP}}{\text{TP} + \text{FN}},
\]

\[
\text{Precision} = \frac{\text{TP}}{\text{TP} + \text{FP}},
\]

\[
\text{F1} = \frac{2 \cdot \text{Precision} \cdot \text{Sensitivity}}{\text{Precision} + \text{Sensitivity}},
\]

where TP represents the number of snoring sounds classified as snoring sounds (true positive), TN is the number of non-snoring sounds truly detected as non-snoring sounds (true negative), FP represents the number of non-snoring sounds falsely recognized as snoring sounds (false positive), and FN is the number of events corresponding to the false detection of snoring sound as non-snoring sound (false negative).

3.4. Classification results

Figure 3 shows the distribution of coefficient 1 and WTCC13 which have the first and second highest coefficients with labels selected by correlation analysis. 5000 samples were randomly selected from 24 subjects in the training set. It shows that the distribution of coefficient 1 and WTCC13 of snoring sounds is different from non-snoring sounds, which could distinguish snoring sounds to a certain extent. Figure 4 shows the overall accuracy of snoring and non-snoring sounds with different wavelet functions under different decomposition layers. The WPT features extracted from different decomposition layers and different Daubechies wavelet functions could work well with accuracies more than 94%.

It can be seen from Fig. 4a that the overall accuracies of WPT features extracted from layer 5 are slightly 0.5 percentage points higher than accuracies...
Fig. 3. The distribution of coefficient 1 and WTCC13 of decomposition layer 5 randomly selected from the training set of 5000 samples. The coefficient 1 and WTCC13 are the coefficient and WTCC of the first sub-band and thirteenth sub-band components. Snoring and non-snoring segments are denoted by red circle and blue circle symbols, respectively.

Fig. 4. The snoring recognition result of two kinds of feature sets under decomposition layer 4, 5, and bark and wavelet function DB1 to DB10 in Daubechies family: a) WPT feature set; b) MFCC and WPT combined feature sets.

of layer 4 under most wavelet functions. With the increase of the decomposition level, the frequency of the signal is divided into smaller bandwidths to obtain much more detailed information about signals, which is beneficial for distinguishing snoring and non-snoring sounds, and yield much better performance. Comparing the recognition rate of layer 5 and bark sub-bands, there is little difference between these two decomposition structures. However, layer 5 and the bark sub-band have the same decomposition construction on low frequencies (0–5.5 kHz), and layer 5 has much finer division than the bark sub-band on high frequencies (5.5–44.1 kHz). In other words, the bark sub-band puts much more emphasis on low frequency. The result shows that the energy of snoring sound and the difference between snoring and non-snoring are mainly on low frequencies. The information on the high frequencies part also cannot be ignored.

It also can be observed from Fig. 4a that the wavelet function also influences the final classification result. The overall accuracies are different between wavelet functions in the same decomposition layer. DB7, DB9, and DB10 in level 5, and DB7 in bark yield much higher recognition for WPT features among all test Daubechies wavelet functions, which are 95.2% approximately, indicating that the wavelet function plays an important role in the decomposition of the signal. Comparing Fig. 4, the WPT and MFCC combined features yielded accuracy with an average value of 95.5%, much better than simple WPT features in terms of overall accuracies under all test wavelet functions and decomposition levels. The difference in recognition results of the MFCC and WPT combined features between decomposition levels and wavelet functions is not as obvious as simple WPT features. The first three recognition rates are DB7 in level 5, DB8 in level 5, and DB9 in L4 which are 95.68%, 95.67%, and 95.65% respectively under all test conditions.

Table 3 shows detailed results for snoring and non-snoring recognition of different feature sets including MFCC, WTCC, WPT features, and WPT+MFCC combined features under the selected DB7, DB8, DB9, and DB10 wavelet functions in decomposition layer 5 and bark sub-bands. It can be known from Table 3 that for WPT features, the accuracy of the three kinds of selected feature sets is comparable, which is around 95.2%. However, the features extracted from DB7-Level 5 yield much higher sensitivity with 96.94%, indicating a higher probability of real snoring sound being recalled. For WPT–MFCC combined feature sets, the features extracted from level 5 with the wavelet function DB7 achieved the best performance during all test conditions considering sensitivity, precision, and F1-score, which are 97.27%, 96.48%, and 96.88%, respectively. Compared with MFCC, simple WPT features, the WPT–MFCC features performed best, and there is an average improvement of 1 percent-
Table 3. Detailed classification results and dimensions of selected features of different feature sets including Cepstral coefficients features, WPT features, and WPT + MFCC combined features under the selected wavelet functions and decomposition layers selected from Fig. 4.

<table>
<thead>
<tr>
<th>Feature-set</th>
<th>Number of features (total)</th>
<th>Accuracy [%]</th>
<th>Sensitivity [%]</th>
<th>Precision [%]</th>
<th>F1 [%]</th>
<th>AUC</th>
</tr>
</thead>
<tbody>
<tr>
<td>Cepstral coefficients</td>
<td></td>
<td></td>
<td></td>
<td></td>
<td></td>
<td></td>
</tr>
<tr>
<td>MFCC</td>
<td>26</td>
<td>94.76</td>
<td>96.58</td>
<td>95.83</td>
<td>96.20</td>
<td>0.98</td>
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<tr>
<td>WTCC</td>
<td>26</td>
<td>93.57</td>
<td>96.17</td>
<td>94.58</td>
<td>95.37</td>
<td>0.98</td>
</tr>
<tr>
<td>WPT</td>
<td></td>
<td></td>
<td></td>
<td></td>
<td></td>
<td></td>
</tr>
<tr>
<td>DB7-L5</td>
<td>69 (256)</td>
<td>95.15</td>
<td>96.94</td>
<td>96.21</td>
<td>96.50</td>
<td>0.99</td>
</tr>
<tr>
<td>DB9-L5</td>
<td>82 (256)</td>
<td>95.15</td>
<td>96.81</td>
<td>96.18</td>
<td>96.49</td>
<td>0.99</td>
</tr>
<tr>
<td>DB10-L5</td>
<td>81 (256)</td>
<td>95.20</td>
<td>96.81</td>
<td>96.23</td>
<td>96.52</td>
<td>0.99</td>
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<tr>
<td>WPT + MFCC</td>
<td></td>
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<td></td>
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<td></td>
</tr>
<tr>
<td>DB7-L5</td>
<td>98 (282)</td>
<td>95.68</td>
<td>97.27</td>
<td>96.48</td>
<td>96.88</td>
<td>0.99</td>
</tr>
<tr>
<td>DB8-L5</td>
<td>96 (282)</td>
<td>95.67</td>
<td>97.44</td>
<td>96.30</td>
<td>96.87</td>
<td>0.99</td>
</tr>
<tr>
<td>DB9-L4</td>
<td>99 (162)</td>
<td>95.65</td>
<td>97.27</td>
<td>96.27</td>
<td>96.77</td>
<td>0.99</td>
</tr>
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</table>

Table 3. Detailed classification results and dimensions of selected features of different feature sets including Cepstral coefficients features, WPT features, and WPT + MFCC combined features under the selected wavelet functions and decomposition layers selected from Fig. 4.

age points and 0.5 percentage points for WPT – MFCC combined features in terms of overall accuracy, recall, precision, and F1-score mentioned in the work. The dimensions of selected features based on the 10-fold cross-validation are also displayed in Table 3. Features with little contribution to the classification are dropped by the selection technology based on correlation coefficients. Based on the aforementioned discussion, the features extracted from the wavelet function DB7 in level 5 decomposition performed better considering simple WPT features and WPT – MFCC combined features with dimensions of 69 and 98, respectively.

We also compared two kinds of cepstral coefficients MFCC and WTCC, which are derived from the Mel-frequency filter and the wavelet packet transform filter respectively. The MFCC outperformed the WTCC with an average improvement of 1 percentage points in terms of evaluation standards mentioned in the work, which means that the Mel-frequency could carry more important information on the upper airway structure variations than wavelet packet transform does.

4. Discussion

In this work, we proposed a novel system to automatically extract snoring sounds from the recorded sounds during sleep based on WPT features. Based on wavelet packet transform, the snoring sound was decomposed into different sub-bands with the same bandwidth and different frequency ranges. Results of WPT features indicated that the information on snoring sounds and the difference between snoring and non-snoring sounds were mainly in the middle and low frequency. With increasing decomposition layer, the signal was decomposed with much smaller sub-bands, and the difference between snoring and non-snoring sounds was much more obviously accompanied by a higher classification accuracy. The snoring sound detection is the first but vital step during the whole analysis system of snoring sounds. Many studies have detected snoring episodes from different kinds of domains.

In previous studies (Han et al., 2006; Karunajeewa et al., 2011; Qian et al., 2015; Solà-Soler et al., 2007; Sun et al., 2022), acoustic features extracted from frequency sub-band of the signal have been demonstrated effectively and widely used in classifying snoring and non-snoring sounds. Qian et al. (2015) used the 1000 Hz sub-band features, and power ratio to detect snoring sound segments. Cavusoglu et al. (2007) explored the sub-band energy distribution of snoring and non-snoring segments by dividing the 0–7500 Hz frequency range into 500 Hz sub-bands, which yielded 90.2% accuracy for simple snorers. These works indicated that the information distribution of snoring sounds is different among sub-bands, which mainly focus on middle and low frequency parts. Moreover, since the bark sub-bands focus on the low frequency part and are sparse in the high frequency part. The WPT furtherly divided the high frequency part based on bark sub-bands which makes the difference in the high frequency much more obviously. And recognition accuracies of layer 5 are slightly higher than bark sub-bands under most wavelet function test conditions. Although the distribution of snoring is mainly concentrated in the middle and low frequency parts, the information of snoring sounds in high frequency part also cannot be ignored. And with increasing of decomposition layer, the difference between snoring and non-snoring sounds is getting obvious, because the signal will be exhibited much more comprehensively by sub-bands with smaller frequency ranges.

The results of works (Cavusoglu et al., 2007; Duckitt et al., 2006; Emoto et al., 2018; Jiang
et al., 2020; Lim et al., 2019; Qian et al., 2015; Sun et al., 2022) showed that MFCC and its related features could yield relatively good performance on recognizing snoring sounds. Jiang et al. (2020) extracted the Mel-spectrogram of signal and used a convolution neural network (CNN) to classify snoring and non-snoring sounds with good performance. Sun et al. (2022) also indicated that different components of MFCC yielded different contributions to the classification result. The same result is also shown in this study. The high effectivity of MFCC and WTCC may be caused by dividing the signal into different sub-bands with the same frequency range through Mel-filter banks based on the mechanism of human hearing.

There are many studies to classify snoring and non-snoring sounds from other parts shown in Table 4 (Adesuyi et al., 2022; Ankişhan, Tuncer, 2017; Arsenali et al., 2018; Jiang et al., 2020; Lim et al., 2019; Nonaka et al., 2016; Sun et al., 2022) used the processing way of images to analyze snoring and non-snoring sound with the recognition accuracy of 95.1% in Mel-spectrogram. In the work of Nonaka et al. (2016), the auditory image model, which has been used to numerically explain the auditory phenomenon of human’s auditory, was developed to automatically extract snoring sounds from sleep sounds, which could achieve a sensitivity of 97.2% from 40 subjects. The work of Lim et al. (2019) and Adesuyi et al. (2022) yielded the highest 99.0% accuracy compared with all other studies. However, these results are not convincing because that there are only 8 and 6 subjects for the two studies. The diversity of samples too small to demonstrate the effectiveness of the proposed algorithm. All these studies are based on subject dependence, which cannot be directly used in practice. It demonstrated evident differences between snoring and non-snoring sounds from all kinds of aspects. The models based on deep learning perform much better than traditional machine learning, which also demonstrates the effectiveness of WPT features proposed in this work. It is worth noticing that these works have yielded competitive results on their own limited dataset. However, there is no sense to compare these accuracies because of the inconsistent dataset used in studies. The dataset of each study is established by its own team with subjects from different counties, different recording equipment, and different labeling standards of snoring and non-snoring sounds. But the result of our work is comparable with previous studies in terms of our own dataset. And it demonstrated that the distribution of snoring and non-snoring sounds in each sub-band of all frequency ranges is obviously different.

In conclusion, there are some contributions to this study. Firstly, it used WPT to extract sub-band features and yielded comparable accuracy in recognizing snoring sounds. Then, it discussed different wavelet functions and decomposition layers, concluding that the difference is getting more evident with the frequency range getting smaller, which is more beneficial for classifying. Thirdly, it demonstrated that although the distribution of snoring sounds is mainly on the low frequency part, there are also differences between snoring and non-snoring sounds in the high frequency part. There are some limitations of the work. The data partition methods mainly included subject dependence and subject independence which greatly influenced the classification performance. Subject dependence is an original data partition method to discuss the features’ influence on classification performance, while the result of subject independence is more suitable for use in practice. In this work, the partition of the training set and test set is based on subject dependence because of the limited subjects. The subject independent classification must base on a huge number of training and validation subjects to make up for the influence of individual characteristics. There are only 24 subjects used in the study, which is hard to perform subject independent classification considering individual characteristics. It is the next step of the paper to collect much more snoring sounds from different subjects to implement detecting snoring sounds based on subject independence.

<table>
<thead>
<tr>
<th>Author</th>
<th>Subjects</th>
<th>Features + classifier</th>
<th>Accuracy [%]</th>
<th>Sensitivity [%]</th>
</tr>
</thead>
<tbody>
<tr>
<td>Jiang et al. (2020)</td>
<td>15</td>
<td>Mel-spectrogram + CNN</td>
<td>95.1</td>
<td>95.4</td>
</tr>
<tr>
<td>Nonaka et al. (2016)</td>
<td>40</td>
<td>Audio image model + MNLR</td>
<td>97.3</td>
<td>97.2</td>
</tr>
<tr>
<td>Sun et al. (2022)</td>
<td>24</td>
<td>Sub-band features + XGBoost</td>
<td>94.3</td>
<td>96.5</td>
</tr>
<tr>
<td>Lim et al. (2019)</td>
<td>8</td>
<td>MFCC, STFT + RNN</td>
<td>98.5</td>
<td>99.3</td>
</tr>
<tr>
<td>Ankişhan, Tuncer (2017)</td>
<td>22</td>
<td>Chaotic features + LLEs</td>
<td>94.4</td>
<td>88.3</td>
</tr>
<tr>
<td>Arsenali et al. (2018)</td>
<td>20</td>
<td>MFCC + RNN</td>
<td>95.0</td>
<td>92.0</td>
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<tr>
<td>Adesuyi et al. (2022)</td>
<td>6</td>
<td>MFCC + CNN</td>
<td>99.0</td>
<td></td>
</tr>
<tr>
<td>This work</td>
<td>24</td>
<td>WPT features + XGBoost</td>
<td>95.15</td>
<td>96.94</td>
</tr>
<tr>
<td></td>
<td></td>
<td>WPT + MFCC + XGBoost</td>
<td>95.68</td>
<td>97.27</td>
</tr>
</tbody>
</table>
5. Conclusion

This study proposed a snoring sounds recognition system based on WPT features and XGBoost classifier. The recorded sleep sounds of 24 subjects, firstly were enhanced and segmented by a subspace noise reduction algorithm and signal presence probability based on the estimation of noise autocorrelation respectively to obtain potential snoring episodes. In the training set, 10-fold cross-validation was implemented to select appreciated features and models. Results of the recognition system showed that features based on sub-bands could well classify snoring and non-snoring sounds with accuracy of 95.65%, sensitivity of 97.27%, and precision of 96.58% in the test set for DB7 function and level 5, the best combination of all test conditions. And the comparison among decomposition layers shows, although the distribution of snoring sounds is mainly in the low frequency part, there is also evident difference between snoring and non-snoring sounds in the high frequency part. However, the MFCC – WPT combined feature set outperformed the simple MFCC and WPT feature sets, with accuracy of 95.68%, sensitivity of 97.27%, and precision of 96.68%. These results have demonstrated that the wavelet packet analysis is effective in recognizing snoring sounds with less computational complexity, which can be further developed to analyze OSAHS at home.

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Ethical approval

This work was approved by the Ethics Committee of Guangzhou Medical University, and an informed consent was obtained from each participant. This work does not contain any studies with animals performed by any of the authors.

References


