

Application of Clustering Techniques for Lung Sounds to Improve Interpretability and Detection of Crackles*

Aplicación de Técnicas de Clustering en Sonidos Adventicios para Mejorar la Interpretabilidad y Detección de Estertores

DOI: <http://dx.doi.org/10.17981/ingecuc.11.1.2015.05>

Research Article - Reception Date: August 19, 2014 - Acceptance Date: March 13, 2015

Germán D. Sosa

Electronic Engineer, Universidad de los Llanos. Villavicencio (Colombia). german.sosa@unillanos.edu.co

Fabián Velásquez Clavijo

Master of Applied Mathematics, Universidad de los Llanos. Villavicencio (Colombia). fvelasquez@unillanos.edu.co

To cite this paper:

G. D. Sosa & F. Velásquez Clavijo, "Application of Clustering Techniques for Lung Sounds to Improve Interpretability and Detection of Crackles," *INGE CUC*, vol. 11, no. 1, pp. 53-62, 2015. DOI: <http://dx.doi.org/10.17981/ingecuc.11.1.2015.05>

Abstract— Due to the subjectivity involved currently in pulmonary auscultation process and its diagnostic to evaluate the condition of respiratory airways, this work pretends to evaluate the performance of clustering algorithms such as k-means and DBSCAN to perform a computational analysis of lung sounds aiming to visualize a representation of such sounds that highlights the presence of crackles and the energy associated with them. In order to achieve that goal, Wavelet analysis techniques were used in contrast to traditional frequency analysis given the similarity between the typical waveform for a crackle and the wavelet sym4. Once the lung sound signal with isolated crackles is obtained, the clustering process groups crackles in regions of high density and provides visualization that might be useful for the diagnostic made by an expert. Evaluation suggests that k-means groups crackle more effective than DBSCAN in terms of generated clusters.

Keywords--- Lung Sounds, Crackles, Vesicular Sounds, Adventitious Sounds, Transformed Wavelet, Decomposition Wavelet, Symlet, Clustering, k-means, DBSCAN, Log-energy.

Resumen— Debido a la subjetividad que involucra actualmente el proceso de auscultación pulmonar y su diagnóstico para evaluar la condición de las vías respiratorias de un paciente, este trabajo busca evaluar el desempeño de los algoritmos de clustering: k-means y DBSCAN para efectuar un análisis computacional de sonidos pulmonares con el objetivo de visualizar una representación de dichos sonidos que exalte la presencia de estertores y la energía contenida en ellos. Para este fin, se emplearon técnicas de descomposición y análisis Wavelet a diferencia del tradicional análisis en frecuencia dada la similitud entre la forma de onda de un estertor típico y la wavelet sym4. Obtenida la señal de sonido pulmonar con estertores aislados, el proceso de clustering agrupa estertores en regiones de alta presencia y ofrece una visualización que puede ser de utilidad para el diagnóstico hecho por un experto. La evaluación hecha sugiere que k-means agrupa conjuntos de estertores de forma más efectiva que DBSCAN en términos de clusters generados.

Palabras Clave— Sonido Pulmonar, Estertores, Sonidos Vesiculares, Sonidos Adventicios, Transformada Wavelet, Descomposición Wavelet, symlet, Clustering, k-means, DBSCAN, log-energia

* Research paper deriving from the research Project "Identificación de sonidos pulmonares mediante uso de fonendoscopia electrónico y procesamiento digital de señales de audio basado en transformada Wavelet". Funded by Universidad de los Llanos y Colciencias. Starting date: July 2013. Ending date: July 2014.

I. INTRODUCTION

Every year around 4 million people die due to lung disease; diseases that affect more than 30 million people around the world, according to studies carried out in 2010 and 2011 [1]. These studies and some others relate the growing number of patients with lung cases to increasingly polluted environments and to social factors such as obesity and the smoking habit.

Given the recurrence of lung disease, the method of auscultation is widely popular due to the rapid diagnosis that provide the ease of the procedure and the stethoscope portability which is an instrument used by the examiner to hear lung sounds, whose identification and characterization is crucial to associate a specific pathology. However, despite the relevance of auscultation to determine the condition of the respiratory system, this procedure has not technologically evolved markedly in recent decades, leaving the nature of the diagnosis entirely on the skills and expertise of the examiner and involving much subjectivity.

A departure point will be the basic classification of lung sounds associated with an abnormal condition known as adventitious sounds (rhonchus, wheezing and crackles). Crackles are particularly common presenting a series of acoustic characteristics that make it difficult to detect, such as its low tone that ranges from 100-200 Hz [2] and a very short time duration ranging between 30-40 ms [3] [4] which permits to describe them according to Laennec, inventor of the stethoscope, as discontinuous sounds of explosive nature and superimposed on the normal lung or vesicular sounds.

Crackles, depending on their frequency within a respiratory cycle, are divided into coarse crackles (CC) and fine crackles (FC), and usually they are due to the presence of secretions or obstructions in the airway. Classifying them as thick (coarse) and thin (fine) provides information about the type and severity of obstruction.

With the use of (more popular today than a few years ago) electronic stethoscopes, it is possible to digitally reveal and capture lung sounds in order to apply some sort of signal processing to help the examiner to facilitate the diagnosis of auscultation. Since then, one of the most frequent approaches in computational analysis of lung sound has been crackling parametrization that allows finding crackles accurately within

a lung sound signal and evaluating their properties, information that can be a support for a diagnosis.

For this purpose, the analysis of data based on Machine Learning has been applied to the analysis of lung sound in order to automatically detect one or more types of sound [5], [6]. To this end, the typical procedure in machine learning involves, on the one hand, a signal preprocessing that emphasizes the defining characteristics for its classification and, on the other hand, a learning phase to develop a discriminative model that identifies each of the concerning sounds.

In the preprocessing stage, because of its short duration, the classical analysis in frequency, normally carried out by the Fourier Transform, is insufficient to capture relevant information about crackling. Moreover, wavelet analysis, thanks to its inherent local analysis of the signal, allows the performance of this task with more reliability than many procedures for signal analysis, as shown in [7], [8], and [9].

As for learning algorithms, this paper proposes an unsupervised learning to detect crackles. Thus forcing the model to detect by itself what portions of sound is observed as abnormal, without the need to compare against the verdict of an examiner. In this way, the robustness of the algorithm does not lie in the amount of crackling sounds noted by an examiner, but the ability of the wavelet to represent crackles satisfactorily.

With the hypothesis that an unsupervised learning provides better detection of crackling given the properties of the wavelet, this article evaluates the performance of the most popular clustering methods (k-means and DBSCAN) within this context and, in turn, it aims at determining whether they can become a diagnostic tool for the medical community support.

II. METHODS

A. Wavelet Analysis

Wavelet analysis, developed by Haar in the early twentieth century and continued by Gabor in 1946 [10], allows multi-resolution analysis of the signal similar to that of the Fourier transform. However, instead of representing the signal in scaled versions of sinusoids, it does so by scaling a signal of short wave called wavelet denoted with the letter Ψ , as shown in (1).

$$WT(x(t)) = \int_{-\infty}^{\infty} x(t)\psi_{a,b}(t)dt \quad \text{with} \quad \psi_{a,b}(t) = \frac{1}{\sqrt{a}}\psi\left(\frac{t-b}{a}\right) \quad (1)$$

Where a and b denote the scaling and movement of the wavelet $\Psi(t)$ and the values contained in WT $(x(t))$ can reconstruct the signal in time $f(t)$.

1) *Discrete Wavelet Transform (DWT)*: In 1988, Mallat [10] developed a new scheme for the wavelet transform, which obtains a subset in powers of 2 (dyadic) of the coefficients of the traditional transform but through digital filters.

This new scheme, allowed a signal to be divided into 2 new samples resulting from a low-pass and a high-pass filter respectively, which are complementary and, when added together, resulted in the original signal.

The product signal from the low-pass filter is called *approach* (A), because it contains the coefficients that contribute most to the waveform of the original signal. Moreover, the result of the high pass filter is called *details* (D) and it contains the remaining coefficients which is information from A that is not contained in the original signal (Fig. 1).

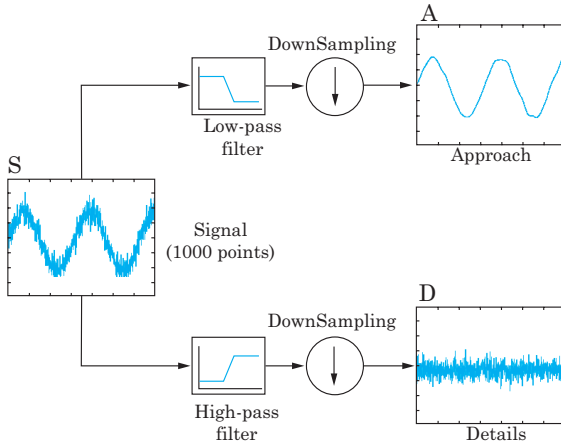


Fig. 1. WDT decomposition stage.
Source : Authors

This decomposition can be performed in cascade, thereby obtaining a decomposition tree that meets at each stage various levels of *approximation* and *details* that together include all information of the original signal (Fig. 2).

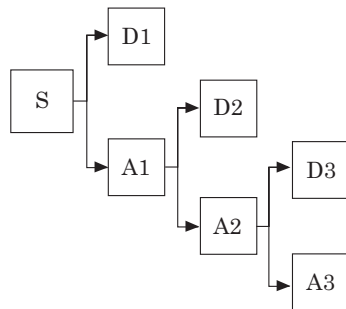


Fig. 2. Three level wavelet decomposition tree.
Source : Authors.

B. Clustering

In the context of machine learning, clustering is one of the approaches of unsupervised learning that aims at grouping subsets called clusters from a dataset, so that the elements belonging to each cluster are the most similar to each other compared to those of other cluster [11].

The concept of clustering in itself is not an algorithm, but the partition task of a dataset into a number of clusters. When mapping the dataset in a vector space of dimension n , where n defines the number of characteristics of the dataset, the clustering process groups the points corresponding to each data in a number of clusters, where each of these groups together data that are "close" within that vector space given some metric, being Euclidean the most common, as shown in (2).

$$\|AB\| = \sqrt{(A,B)} \quad (2)$$

Currently there are a variety of clustering algorithms that are widely used. However, given the typical properties of a lung sound signal, not all of them offer an efficient performance. Such characteristics require an algorithm that is scalable in the order of several thousand points presenting a vector with recorded audio quality (higher sampling frequency to 10 KHz), likewise, the number of clusters, corresponding to found crackling groups can vary from a few or dozens. Therefore, given the recommendations of machine learning module scikit-learn [11], this work implements those clustering algorithms that have greater scalability, once provided the amount of points and clusters, which are k-means and DBSCAN.

1. *k-means*: The k-means algorithm groups data into clusters seeking to separate dataset in k groups presenting the same variance of the data assigned to each cluster in connection to the center of mass of the same, called centroid [12]. Therefore, the algorithm finds k centroids so as to minimize the expression (3).

$$J(X,C) = \sum_j^k \min_{\mu_i \in C} \|x_j - \mu_j\|^2$$

$$\text{con } j = 1, 2, 3 \dots \text{ number of samples} \quad (3)$$

2. *DBSCAN*: A different way to approach the task of grouping data into clusters is performed by DBSCAN. This algorithm sees clusters as high density areas separated by areas of low density [11]. Under this procedure so generic, DBSCAN clusters may have any shape, rather than regular clusters of k-means, which for example, using an L^2 norm tend to be circular.

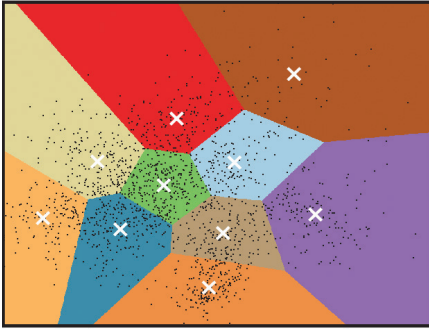


Fig. 3. k-means applied to a 2D dataset.

Source: <http://scikit-learn.org/stable/modules/clustering.html> [11]

More formally, the DBSCAN algorithm is centered on the concept of core-samples, which are samples in high-density regions of the dataset that are close to each other under any metric. Likewise, there are the non-core-samples, which oppositely, are samples in low-density regions that are close to a core-sample but are not core-samples themselves.

Given these concepts, DBSCAN define a cluster by means of two parameters, a distance d and a minimum of m samples. Having m samples of the dataset, that all meet the fact of being separated from one another by a distance less than or equal to d ; this subset of samples, will define a cluster.

Regarding k-means, DBSCAN has the advantage that the number of clusters is not defined by the user and that not all elements of the dataset are assigned to a cluster and can be discarded and considered due to being located in areas of low density. However, a potential drawback is that it is very sensitive to variations of the parameters d and m . When trying the wrong parameters, the algorithm can define too many or too few clusters for the given dataset.

Fig. 4 shows an example of DBSCAN on a 2D dataset, the black points are those considered in areas of low density.

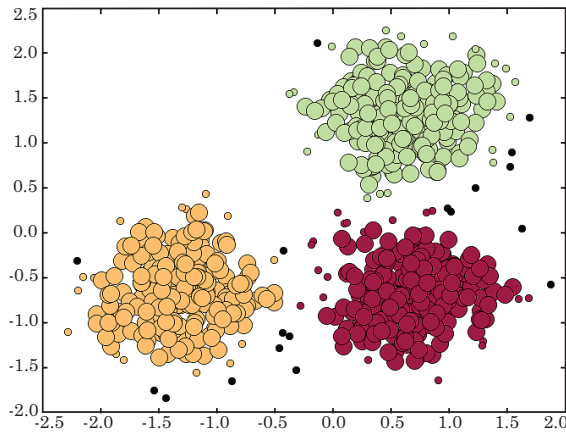


Fig. 4. DBSCAN applied to 2D dataset. In this example, three clusters are demarcated based on a defined m and d .

Source: <http://scikit-learn.org/stable/modules/clustering.html> [11]

C. Databases

Lung sounds, on which the performance of the procedures described in Section IV was evaluated, correspond to 20 audio samples from RALE Repository [13], which corresponds to lung sounds with presence of coarse or fine crackles.

Each sound has a sampling frequency $F_s = 10240$ Hz. In addition, each sample was converted to WAV format with 16-bit PCM modulation to a single channel using the Audacity [14] software and cropped so that it keeps the corresponding sound of a respiratory cycle.

III. DESIGN

A. Removing Crackles by Wavelet Transform

Former works for separating vesicular sounds from crackling sounds by wavelet transform were done in [7] and [8]. A similar procedure was conducted for the above, as shown below:

A three level wavelet decomposition tree was implemented in MATLAB R2011b language, using Symlet wavelet family, more specifically, the wavelet sym4.

This wavelet was chosen because of its similarity in waveform with a typical crackle, as shown in Fig. 5.

1) *Reconstruction of signal and thresholding*: once decomposition is performed on the signal, it is necessary to rebuild so that only relevant information is retrieved.

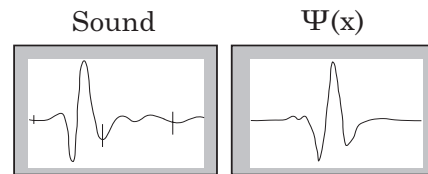


Fig. 5. Typical crackle waveform (right); Wavelet sym4 (left).

Source: "Elimination of vesicular sounds from pulmonary crackle waveforms" Computer methods and programs in biomedicine (2008) [3].

Taking a similar scheme to that of Fig. 2, each new stage of decomposition comes from breaking down each *approach* to the three levels established.

Provided the assumption that the crackles behave as *details* on the decomposition, each level extracts different types of crackles, from the least scaled in time (D1) to the shortest duration (D3).

A threshold stage is used for eliminating coefficients in each level which are very small in relation to those extreme values of the signal. These atypical extreme values are those which occur due to the appearance of a crackle [8]. Values located within the range of quartiles Q1 and Q3 are eliminated using the median as the estimator.

Once these new coefficients are obtained for details, the signal is reconstructed inversely to the decomposition process, although for this procedure, the last level of approximation is not used (see Fig. 6); the results are as shown in Fig. 7.

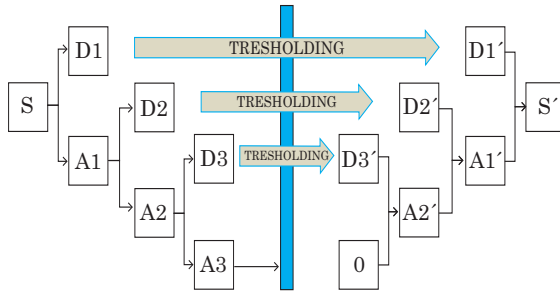


Fig. 6. Scheme of decomposition, threshold and reconstruction. D1, D2 and D3 containing the high frequency coefficients of the signal S. A threshold is implemented in each one for obtaining D1', D2' and D3'. The new signal S' is reconstructed from these new details regardless A1, A2 or A3.
Source: Authors

B. Clustering over Extracted Crackles

Samples of lung sound signals converted into time, where crackles can be highlighted using wavelet transform over the rest of the sound, can be of great help to the interpretability in auscultation.

However, to effectively group subsets found crackles and see what energy they contain within the signal is a task that can hardly be made by an examiner and, if he had this information at hand, this would further support the diagnosis of the patient. Therefore, clustering algorithms are arranged in a form suitable for this procedure, and using the results obtained, it is possible to make a visual representation of these for interpretation.

Due to the nature of audio signals, consisting of a 1D vector of tens of thousands of samples, as a typical audio signal, not all clustering algorithm scales well for the task to do. For this work, k-means and DBSCAN were chosen because they scale well for large volumes of data, provided they do not have too many dimensions and, in the case of k-means, the number of clusters is not high.

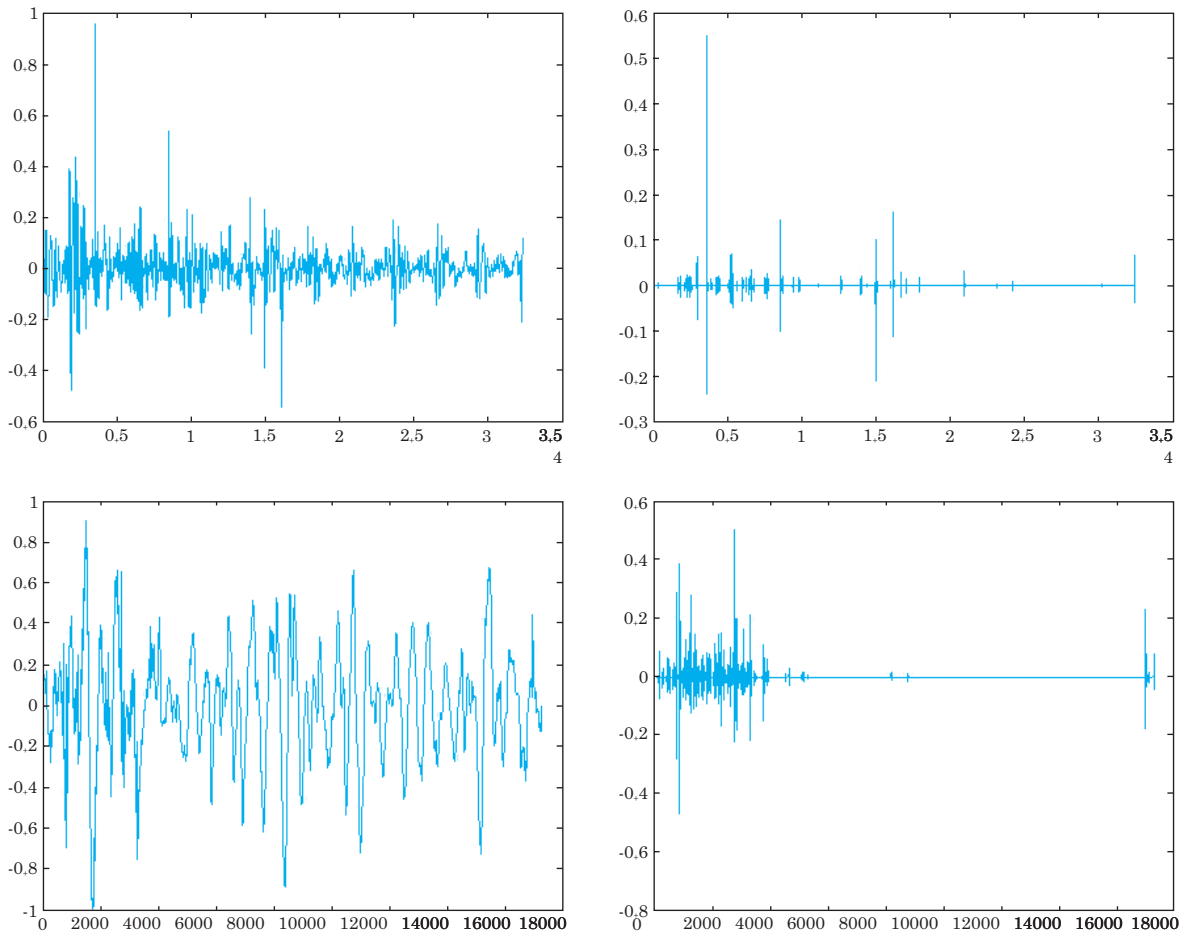


Fig. 7. Signs of lung crackling sound (left); reconstructed signals from the details of wavelet decomposition tree (right).

Source: Authors

The clustering procedure was implemented in Python 2.7.3 using IPython programming environment. Additionally, clustering algorithms were provided by the module scikit-learn v0.14 [11], designed to work in this language.

1) *Preprocessing (crackling indexing)*. Time signal S (see Fig 6) provided by the wavelet analysis is not useful for the clustering procedure as it does not directly represent in what instant of time a crackle appears. This signal corresponds to a series of values of the same length as the original signal S which emphasizes the presence of crackles and sets the rest to 0.

Information relevant to the process of clustering is the location of the crackling, so a new vector S_{pos} is created from the crackling signal S' that stores indexes S' that are due to the presence of a crackle. That is, $S_{pos} \text{ indexes} = |S'| > 0$.

2) *Clustering with k-means*: once obtained the vector with the location of each crackle, the k-means algorithm from scikit-learn module [11] is applied to the S_{pos} vector.

The only relevant k-means parameter is the number of clusters k . There are other user-adjustable parameters, like how to initialize the cluster or the number of iterations, but these are focused on managing the computational cost.

To set the number of clusters k , an elbow curve was used to determine optimal k manually. The elbow diagram shows the inertia of implementing k-means based on k . The inertia is expressed as shown in (4).

$$Inertia = \sum_{j=1}^k \sum_{i=1}^l \|x_i - \mu_j\|^2 \text{ to } l \text{ samples (4)}$$

Logically, each elbow curve is different for each sound. However, all comply with a similar pattern where the inertia goes down to almost 0 using over 10 kernels as shown in Fig. 8.

3) *Clustering with DBSCAN*: As in k-means, DBSCAN is applied to the index crackling vector S_{pos} to estimate the optimal value of its parameters.

DBSCAN requires the parameters d and m (see section II) to adjust the model on a dataset. In this case, d was fixed as = 1000, as it was assumed that separated crackles among themselves by a space of 100 ms or less are due to a high density area. Given the sampling frequency $F_s = 10240$ Hz, it means that 100 ms comply with a space of $F_s * 0.1 \approx 1000$ points in the time series.

With fixed d , it is easier to determine the optimal value of m by a curve that plots the number of clusters found by DBSCAN depending on m .

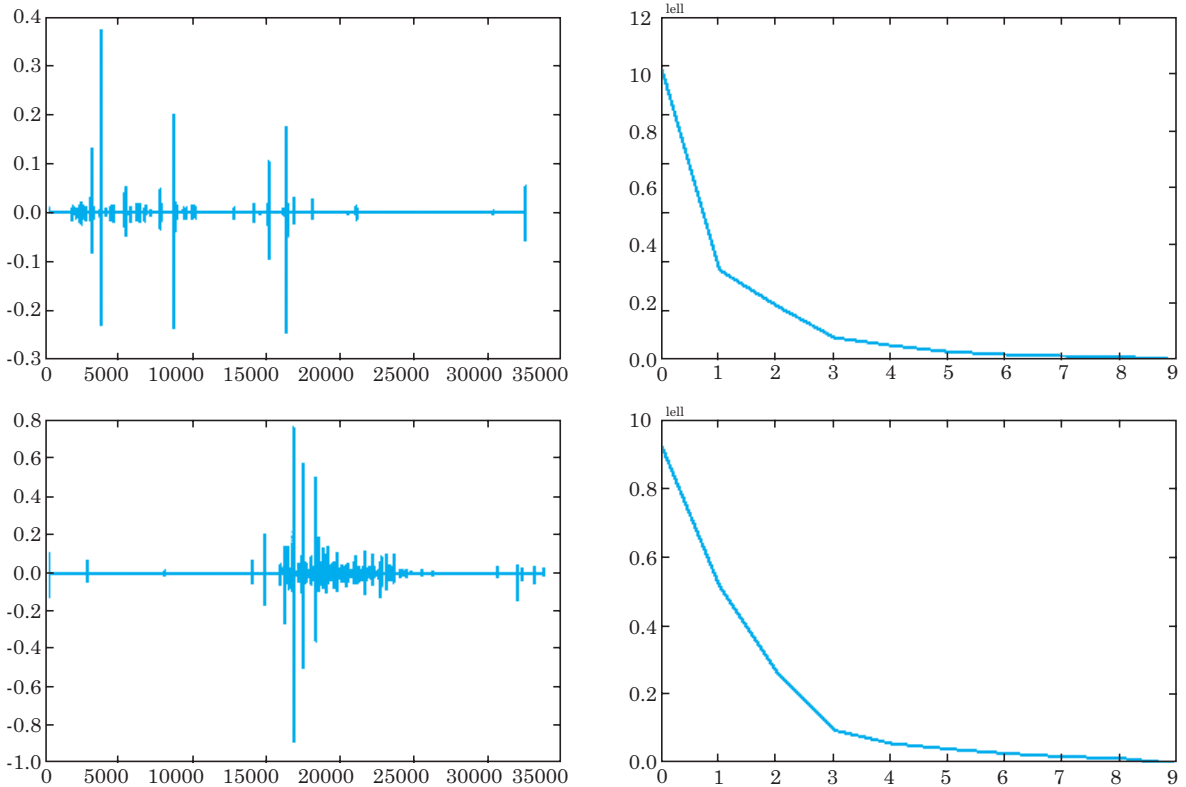


Fig. 8. elbow curves (right) for two signal samples with extracted crackling (left). The minimum inertia is achieved with $k > 10$.

Source: Authors

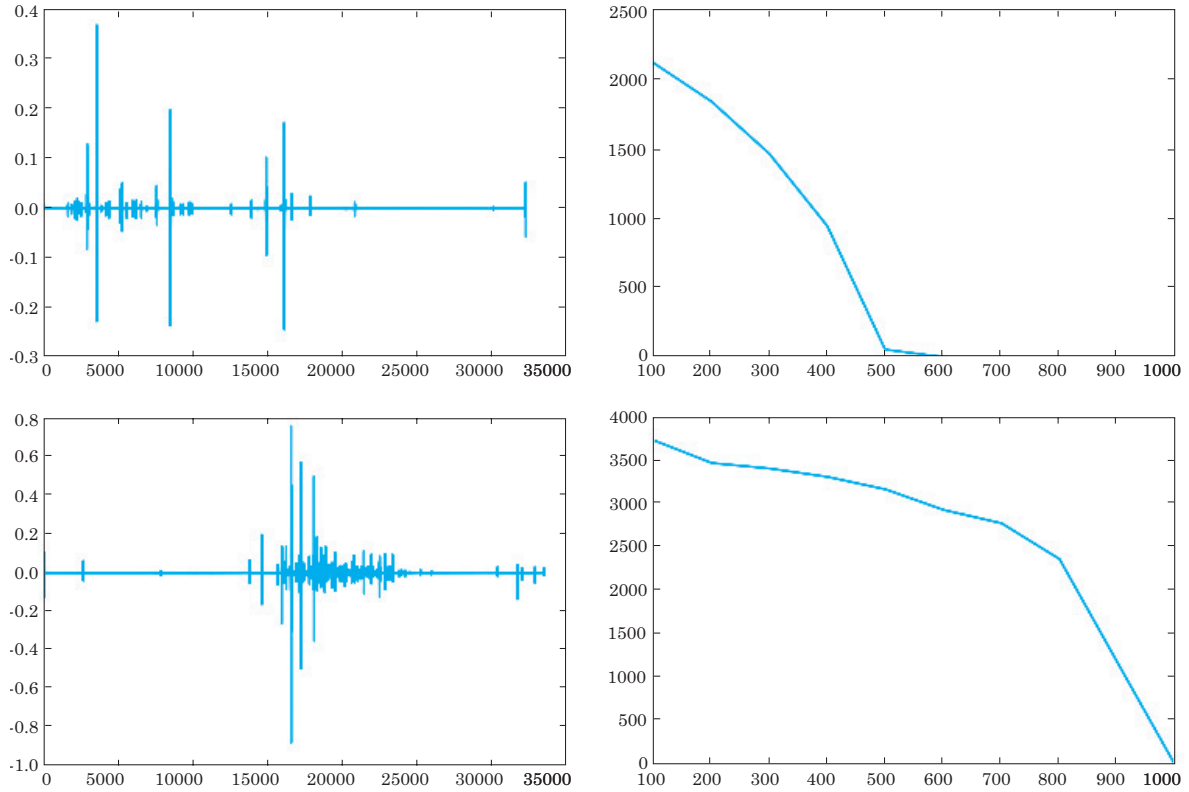


Fig. 9. Curves showing the number of clusters found by DBSCAN depending on the number of core-samples needed to define a cluster (right) corresponding to the signal samples with accented crackles of each curve (left).

Source: Authors

It is important to note that when $m > d$, the number of clusters is 0. This case can only be present in this type of one-dimensional signals, because if the algorithm DBSCAN is asked to find more than m samples in a space of k points in one dimension, it is logically impossible, and therefore the entire sample will be considered as non-core-samples.

Based on the results of Fig. 9, together with those obtained from the other samples, it is difficult to determine a fixed value for m for the entire dataset, because when m exceeds a certain value, the number of found clusters drops steeply until almost 0. However, in order to fix a value, $m = 200$ is established, which defines a consistent number of clusters, although very high compared to k-means.

IV. RESULTS

K-means and also DBSCAN were applied to a set of twenty pulmonary signals with crackling sounds. The information on each one of the found clusters was used to create a visual representation, as shown in Fig. 10, which draws every cluster in a time window equal in length to the original signal.

An important observation about these graphs is that unlike k-means where the clustering number

is not greater than 10, the abundant number of clusters estimated by DBSCAN generates a graph that does not differ much from that obtained if we equal any found value in the crackling signal S 'to a unique value, for example, to 1.

Therefore, it was decided to establish k-means as $k = 10$ clusters for displaying crackles of a sample of lung sound.

A. Measurement of Post-clustering Energy

To display the partitions obtained by clustering has the disadvantage that it does not take into account some information relating to the energy contained in the S_{pos} crackling signal, as only their location is taken into account for the clustering process.

That's why as a final addition to the visualization, the color of each partition process regarding the energy contained in each cluster of the measured signal on the portion between t_1 and t_2 corresponding to the signal S as in (5) is determined.

$$E_s|_{t_1}^{t_2} = \sum_{i=t_1}^{t_2} |x[i]|^2 \quad (5)$$

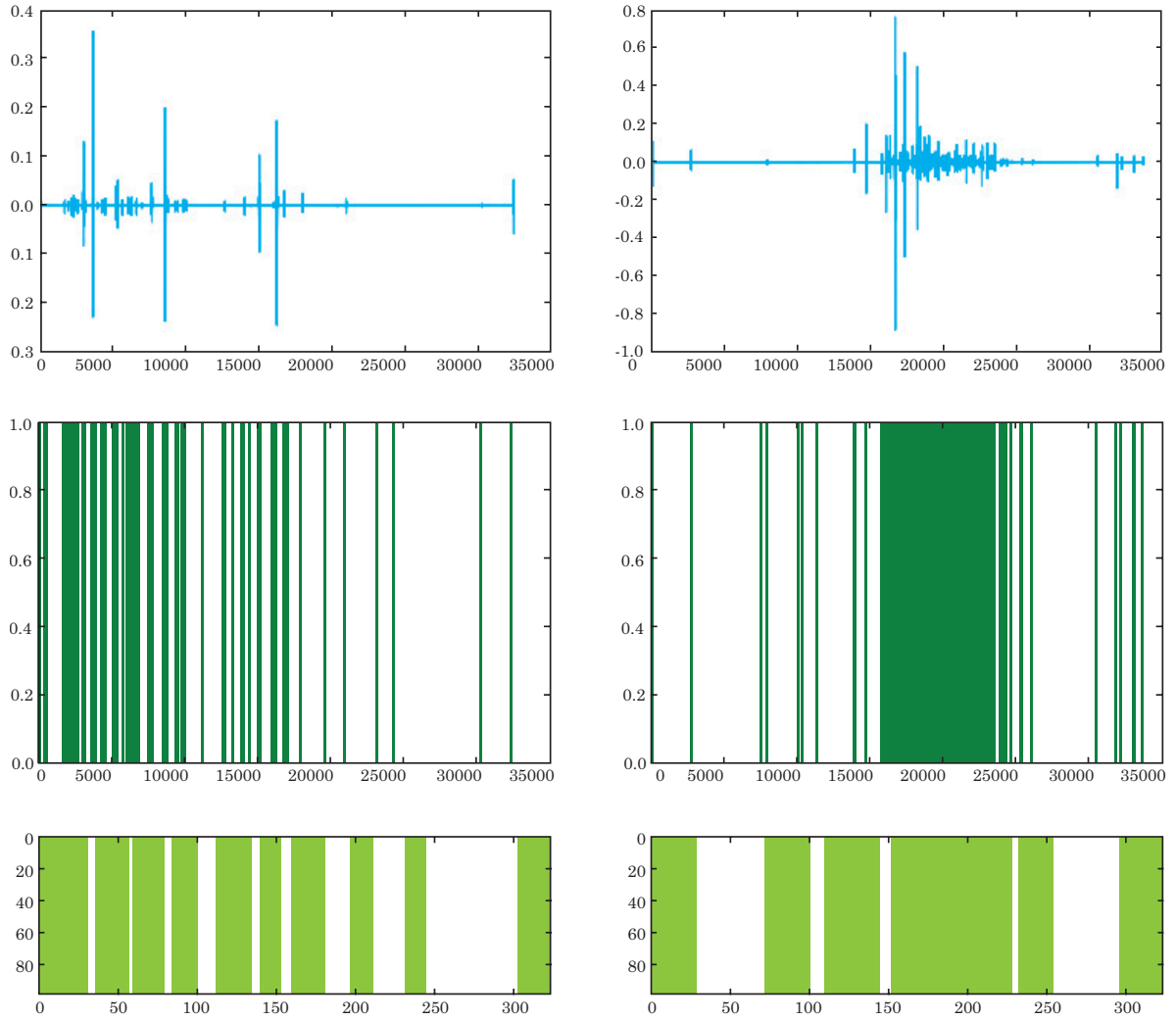


Fig. 10. Display of clusters obtained by DBSCAN and k-means (2nd and 3rd row respectively).
Source: Authors

Due to low amplitude of the S_{pos} signal, the energy measured by the above equation results in very small values. Therefore, a slight change was made in a base 10 logarithmic scale for easy interpretation as shown in (6).

$$E'_s = \frac{10 + \log_{10}(E_s)}{10} \quad (6)$$

Using the values of E_0 s for coloring each cluster, a new presentation is obtained, as shown in Fig. 11.

V. ACKNOWLEDGMENTS

D. Germán Sosa thanks for their support and funding to the Young Scholars Program No.

566/2012 of the Administrative Department of Science, Technology and Innovation COLCIENCIAS and Research Institute of the Colombian Orinoco IIOC of Universidad de los Llanos respectively.

VI. CONCLUSIONS

The proposal for a visual representation by making use of techniques of wavelet analysis and clustering that facilitates the interpretation of how crackles affect lung sound is shown possible and with results that can support a medical diagnosis when linking found patterns in these visualizations with a specific pathology.

Of the various clustering algorithms currently used, there were chosen those known as scaling

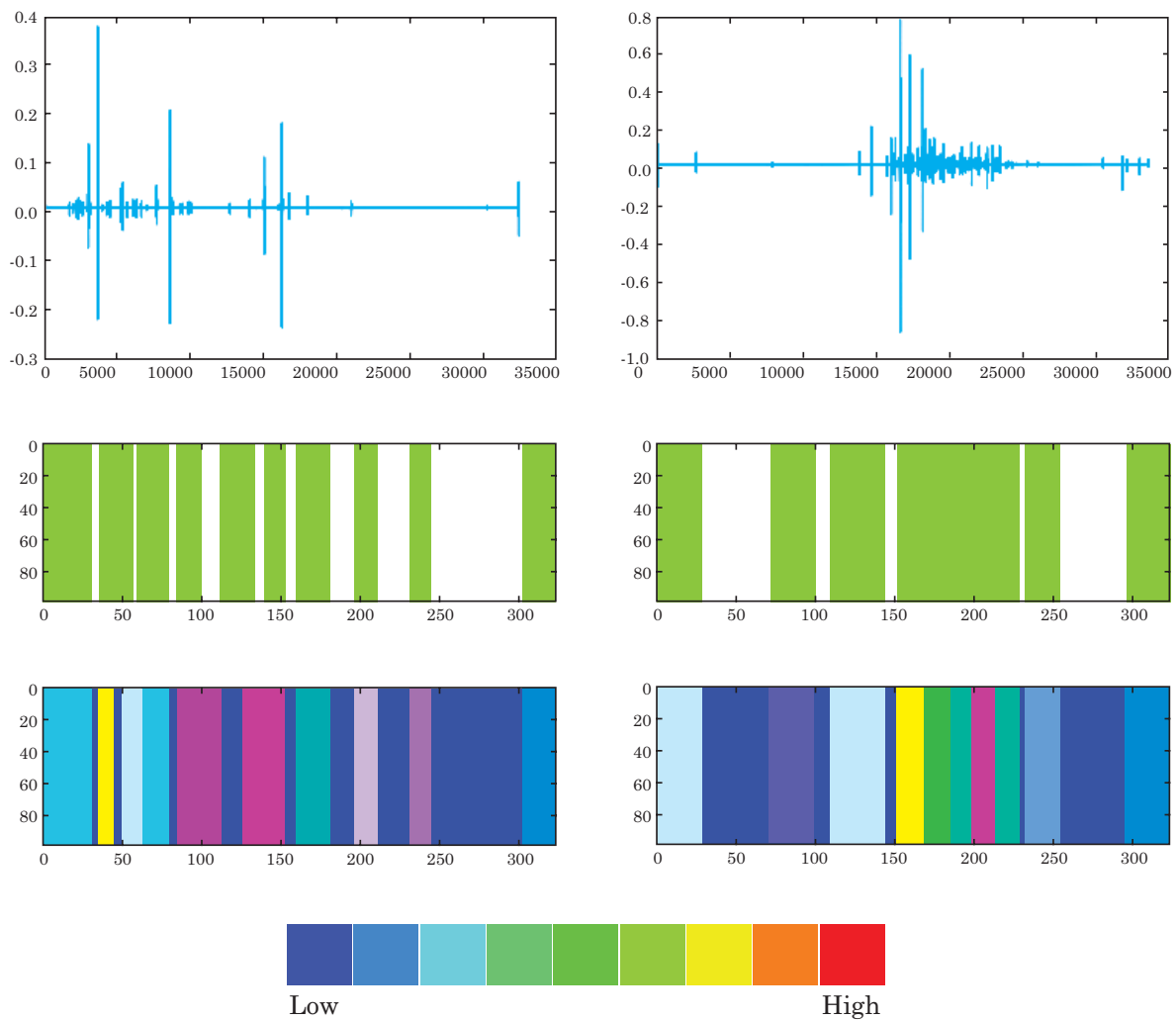


Fig. 11. Display of clusters obtained by k-means (2nd row) and color clusters according to the energy contained in S (3rd row).
Source: Authors.

well with large dataset and not many clusters (k-means and DBSCAN), which complies with the nature of an audio signal and the aim sought in this work. However, due to the instability of the m parameter for the DBSCAN algorithm and the huge amount of estimated clusters, DBSCAN was discarded leaving k-means with $k = 10$ clusters as partitioning method of the signal in representative groups of crackling.

The graphs obtained after measuring the energy in each partition facilitates the interpretation of how the presence of crackles affect a lung sound signal specifically. Even, due to the little information that is required to obtain this chart (location of the centroid, width of each cluster, and corresponding energy), it is possible that this representation will work for a lung sound classifier

based on machine learning, however, this hypothesis should be tested in future work.

REFERENCES

- [1] W. H. Organization et al., "The top 10 causes of death: The 10 leading causes of death in the world, 2000 and 2011," 2013.
- [2] M. Munakata, H. Ukita, I. Doi, Y. Ohtsuka, Y. Masaki, Y. Homma, and Y. Kawakami, "Spectral and waveform characteristics of fine and coarse crackles," *Thorax*, vol. 46, no. 9, pp. 651–657, 1991. DOI:10.1136/thx.46.9.651
- [3] M. Yeginer and Y. P. Kahya, "Elimination of vesicular sounds from pulmonary crackle waveforms," *Computer methods and programs in biomedicine*, vol. 89, no. 1, pp. 1–13, 2008. DOI:10.1016/j.cmpb.2007.10.002
- [4] T. Kaisia, A. Sovijarvi, P. Piirila, H. Rajala, S. Haltsonen, and T. Rosqvist, "Validated method for automatic detection of lung sound crackles," *Medical and Biological Engineering and Computing*, vol. 29, no. 5, pp. 517–521, 1991. DOI:10.1007/BF02442324

APPLICATION OF CLUSTERING TECHNIQUES FOR LUNG SOUNDS TO IMPROVE INTERPRETABILITY AND DETECTION OF CRACKLES

- [5] M. Yeginer and Y. P. Kahya, "Probing the existence of medium pulmonary crackles via model-based clustering," *Computers in biology and medicine*, vol. 40, no. 9, pp. 765–774, 2010. DOI:10.1016/j.combiomed.2010.07.004
- [6] M. Paoletti, G. Camiciottoli, E. Meoni, F. Bigazzi, L. Cestelli, M. Pistolesi, and C. Marchesi, "Explorative data analysis techniques and unsupervised clustering methods to support clinical assessment of chronic obstructive pulmonary disease (COPD) phenotypes," *Journal of biomedical informatics*, vol. 42, no. 6, pp. 1013–1021, 2009. DOI:10.1016/j.jbi.2009.05.008
- [7] X. Lu and M. Bahoura, "An automatic system for crackles detection and classification," in *Electrical and Computer Engineering, 2006. CCECE'06. Canadian Conference. IEEE, 2006*, pp. 725–729. DOI:10.1109/CCECE.2006.277698
- [8] M. Bahoura and X. Lu, "Separation of crackles from vesicular sounds using wavelet packet transform," in *Acoustics, Speech and Signal Processing, 2006. ICASSP 2006 Proceedings. 2006 IEEE International Conference*, vol. 2. IEEE, 2006, pp. II–II. DOI:10.1109/ICASSP.2006.1660533
- [9] L. J. Hadjileontiadis and S. M. Panas, "Separation of discontinuous adventitious sounds from vesicular sounds using a wavelet-based filter," *Biomedical Engineering, IEEE Transactions on*, vol. 44, no. 12, pp. 1269–1281, 1997. DOI:10.1109/10.649999
- [10] S. Mallat, *A wavelet tour of signal processing*. Academic Press, 1999.
- [11] F. Pedregosa et al., "Scikit-learn: Machine learning in Python," *J Mach Learn Res*, vol. 12, pp. 2825–2830, 2011.
- [12] E. Alpaydin, *Introduction to machine learning*. MIT press, 2004.
- [13] J. J. Ward, "Rale lung sounds 3.1 professional edition," *Respiratory Care*, vol. 50, no. 10, pp. 1385–1388, 2005.
- [14] D. Mazzoni, M. Brubeck, and J. Haberman, "Audacity: Free audio editor and recorder". [En línea] Disponible en: <http://audacity.sourceforge.net>, 2005.