MATHEMATICAL MODEL OF VOCAL SIGNALS FOR THE TASKS OF HUMAN VOCAL APPARATUS DIAGNOSTIC

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Abstract. The necessity of voiced fricative sounds processing is grounded for the problem of human vocal apparatus diagnosis. The analysis of possibility of fricative sound modeling as the determined and stationary random process is conducted. The inadequacy of these models is found out for the task of vocal organs diagnostics on the early stages of disease, because they do not take into account the signal periodicity, eventuality of forms of pronunciation violation, related to the functional state of vocal organs. The mathematical model of fricative sounds as the periodically correlated stochastic process is grounded with the use of energy theory of stochastic signals, which takes into account a time structure, combination of eventuality and periodicity of signal.

Keywords: vocal signals; fricative sound; periodically correlated stochastic process; human vocal apparatus; diagnostic.

Introduction. According to statistics of Ministry of Health of Ukraine and the World Health Organization, every year there is a tendency to increase the number of people with diseases of the vocal apparatus organs. Therefore, an important task of modern medicine is timely diagnosis of pathological changes in the vocal apparatus in the early stages of their emergence and development.

Pathological changes of the vocal apparatus lead to irregularities in their work [1]. It finds a clear reflection in voice signals - sonorant consonants, due to the specifics of their creation. Early diagnosis allows to detect changes in the functional state of the vocal apparatus by proper processing of vocal signals and to conduct preventive measures or to choose medical treatment. For an objective diagnosis in medicine indirect methods are used, based on system-signal concept, according to which the voice signal is treated as a physical process that extends from the object and is a means of transferring information about this object. The efficiency of the diagnostic system functioning is crucially determined by the mathematical model of

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the signal, that is at its basis, and shall include in its structure an informative signs of changes in the vocal apparatus. It is necessary to substantiate the algorithms of measurement and processing characteristics of voice signals and interpreting the results.

The actual problem is the grounding of voice sounds mathematical model and development of methods of their analysis for automated computer diagnostic systems, oriented to the problem of early diagnosis of human vocal apparatus functional state by introducing a new class of informative signs, that are indicators of pathological conditions of the human vocal apparatus.

Materials and Methods. Mathematical model of the vocal sound signals should be adequate as research problems and their physical nature and structure. For the diagnosing problems it is necessary to process the class of voice signals, which are the most sensitive to changes in the functional state of the vocal apparatus. According to the logopedic statistics [1-3], in the presence of disruption in the work of vocal tract, the irregularities in the pronunciation of voiced fricative sounds (VFS) are observed. For the research the sound [*l*] was selected. The creation of such sound utilized almost all organs of human vocal apparatus. So, this voice signal contains much information about the work of vocal apparatus.

The process of VFS creating by humans vocal apparatus was analyzed [4,5]. With the creation of VFS in the flow of exhaled air (Fig. 1 (1)) source of signals form the sound of signals with a typical repeatability - the basic tone (Fig. 1 (3)) generated by the vocal folds, which are excited quasi-periodic sequence of nerve impulses p(t) (Fig. 1 (2), Fig. 2). Articulation apparatus forms the phonetic structure of signals x(t) (Fig. 1 (4), Fig. 2). So voice signal y(t) can be represented as a complex amplitude modulated signal $y(t)=p(t)\cdot x(t)$, where y(t) - signals VFS (sound [I], shown in Fig. 2) - message; p(t) - carrying signal, characterizing the work of signal source; x(t) - the envelope of the voice signals in the time domain, which describes the behavior of the speech apparatus in time. Analysis of envelope and carrier signals of VFS in the time and frequency domain will enable estimate of the source of signals and speech apparatus as a whole and its organs in particular. Accordingly mathematical model must take into account the frequency, caused by the vocal folds work, and have a means of analyzing time-phase structure, that reflects the behavior of the source of the source of signals.

Grounding the VFS mathematical model. Fig. 3 shows the experimentally registered sounds [1] from the patients in a state of medical norm and pathology.

Due the requirements of the VFS mathematical model, defined physical nature and task of diagnostic, namely, taking into account the such signals frequency and its time-phase structure, the simplest model of VFS is the deterministic model as a mixture of periodic functions [6]. Informative and invariant

signs of the signal are the morphological features of signals time structure and amplitude spectra, evaluation of which characteristics underlie the methods of harmonic and formant analysis.

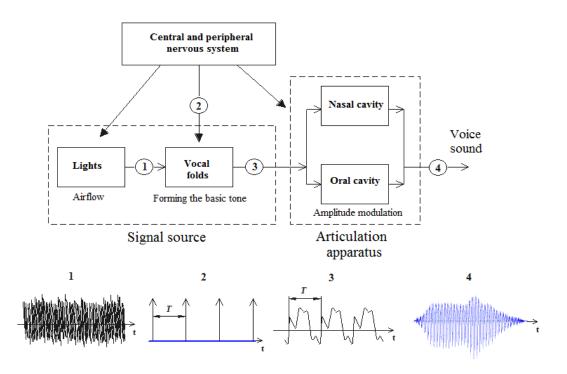


Fig. 1. The process of VFS creating

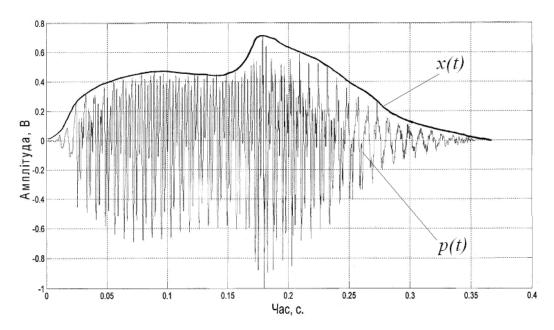


Fig. 2. VFS, as a result of the speech formation process

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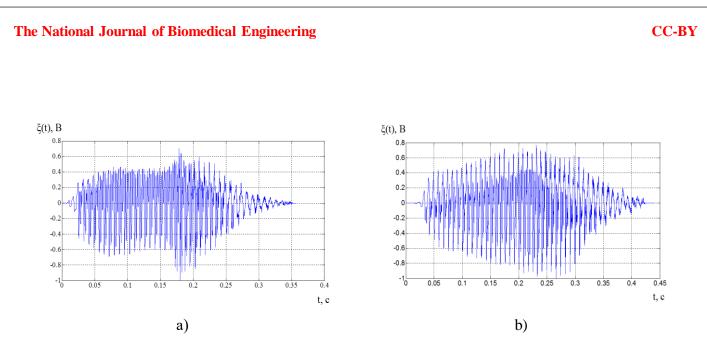


Fig. 3. Experimentally registered sounds [l] (a) - the state of the vocal tract is normal; (b) - the state of the vocal tract is pathology

The estimates of amplitude spectra for the same volume of samples, taken from VFS signals are shown in Fig. 4.

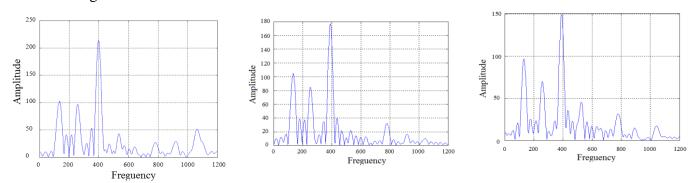


Fig. 4. Estimates of amplitude spectra of samples from experimentally registered sounds sound [*l*]. Duration sample - 0.03 sec

Results. The results of VFS harmonic analysis methods in presenting them as deterministic periodic process confirm that the resulting amplitude spectra VFS reviews (Fig. 4) are variable. This fact indicates the presence of stochastic component in such signals.

In the case of probabilistic approach to modeling of VFZ, the mathematical model of such signals in the form of stationary stochastic process is known [6], which defines the methods of spectral-correlation analysis. In this case, the informative and invariant features of its signal are the probability characteristics

and their distributions. This mathematical model reflects the complexity of VFS in the spectral power distribution, but it does not reflect the time-phase structure, which is important to identify the time moments of occurrence of changes in the vocal apparatus functioning In terms of energy theory of stochastic signals [7] these requirements satisfies a model in the form of periodically correlated stochastic process (PCSP), which has a means to take account of connectedness the signal frequency and probability of its characteristics changes.

PCSP of class π^{T} – is the process, which mathematical expectation is the periodic function and its period is equal to the period of process correlation, and the correlation function satisfies the conditions [7]:

1) periodicity according to the shifts on the period of correlation $r_{\xi}(t+T,s+T) = r_{\xi}(t,s)$, T > 0 for all $t, s \in \mathbf{R}$;

2) the middle power is finite during this period $\frac{1}{T}\int_{0}^{T}r_{\xi}(t,t)dt < \infty$.

Image the stochastic oscillating process (SOP) as PCSP should submit it via stationary components [7]

$$\xi(t) = \sum_{k \in \mathbf{Z}} \xi_k(t) \cdot e^{ik \frac{2\pi}{T}t}, t \in \mathbf{R} , \qquad (1)$$

where $\xi_k(t)$ – stationary and stationary associated components of SOP in presenting it as PCSP; Z – the set of all integers; T – signal period of correlation.

Components $e^{ik\frac{2\pi}{T}t}$ of expression (1) reflect the harmonious (fluctuating) structure of signals, and components $\xi_k(t)$ – its stochasticity.

However, the analysis of temporal structure of VFS shows that they are close in appearance to amplitude modulated oscillations. In this case, using the concept of beating from the vibrations theory, VFS advisable to interpreted as poly-PCSP, and as the most important for applications – bi- PCSP. Bi-PCSP has image [7]:

$$\xi(t) = \sum_{k,j \in \mathbb{Z}} \xi_{kj}(t) e^{i\lambda_{kj}t},$$
(2)

$$\lambda_{kj} = k\Lambda_0 + j\Lambda, \ \Lambda_0 = 2\pi/T_0, \ \Lambda = 2\pi/T, \ k, j \in \mathbb{Z},$$

where T_0 , T – the periods of correlation.

However, there are two ways to image bi-PCSP by the periodically correlated components:

1)
$$\xi(t) = \sum_{k \in \mathbb{Z}} \xi_k^{(T)}(t) e^{ik\Lambda_0 t}$$
, $\xi_k^{(T)}$, $k \in \mathbb{Z}$ – bi- periodically correlated stochastic processes

with the same period correlation $T = 2\pi / \Lambda$;

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2)
$$\xi(t) = \sum_{j \in \mathbb{Z}} \xi_j^{(T_0)}(t) e^{ij\Lambda t}$$
 through periodically correlated components $\xi_j^{(T_0)}$, $j \in \mathbb{Z}$, with

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common correlation period $T_0 = 2\pi / \Lambda_0$. In this way the reduction of order of stochastic processes correlation and construction bi-PCSP to PCSP is implemented.

So as the model of VFS is used PCSP as a result of lowering the order of correlation, which features VFS envelope in the time domain will be included in the characteristics of its carrier.

The model of VFS in the form of PCSP defines the common methods of statistical processing of signals: synphase and component, and is used to calculate the statistical assessments of their probability characteristics, which in the case of VFS processing are informative signs and indicators of the human vocal apparatus state.

Discussion. The analysis of the VFS structure and described properties of periodically correlated stochastic processes implies that the mathematical model of such class of processes makes it possible to adequately describe the signal of VFS, namely to consider a combination the stochasticity and periodicity of signals, and therefore develop the methods for determining invariant informative signs of sound, based on statistics of such signals for the task of diagnosis of human vocal apparatus diseases.

Conflict of interest statement: The authors state that there are no conflicts of interest regarding the publication of this article.

Author Contribution: Conceptualization: Vasyl Dozorskyy. Data curation: Vasyl Dozorskyy. Formal analysis: Vasyl Dozorskyy. Writing - original draft: Oksana Dozorska. Writing - review & editing: Leonid Dediv. 1. Jafek B, Stark A. ENT secrets. Philadelphia, PA: Hanley & Belfus; 1995.

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