

## SIGNAL DENOISING USING A LOW COMPUTATIONAL TRANSLATION-INVARIANT-LIKE STRATEGY INVOLVING MULTIPLE WAVELET BASES: APPLICATION TO SYNTHETIC AND ECG SIGNALS

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### Abstract

In this paper, an efficient method for the denoising of *electrocardiogram* (ECG) signals is presented. As it is well-known, the efficient *translation-invariant* (TI) denoising technique, first introduced by Coifman and Donoho, uses  $K$  pre-processing shift-rotation operations,  $K$  denoising operations similar to the standard Donoho's thresholding algorithm,  $K$  post-processing inverse shift-rotation operations, and finally, the  $K$  new less noisy copies generated by the preceding steps are averaged to produce a final denoised signal. Thus and conversely to the previously mentioned TI algorithm, the suggested technique consists of the design of a low computational translation-invariant-like strategy that eliminates the  $K$  pre-processing shift-rotation and the  $K$  post-processing inverse shift-rotation operations and only keeps the  $K$  wavelet-based denoising operations where for each one we use a different mother wave among a set of  $K$  mother waves  $\psi_1, \psi_2, \dots, \psi_K$ . Consequently, each mother wave generates a new less noisy copy from the original noisy signal. Finally, the produced less noisy multiple copies are averaged to reach the final denoised signal. Through this strategy, we can avoid the use of multiple hardware sensors to generate multiple noisy copies to be averaged to restore the clean version of the signal. Consequently, the proposed approach can considerably reduce the cost of the acquisition system. Additionally, the several results produced from extensive simulations show that the proposed algorithm outperforms many translation-invariant-like methods and can be considered as one of the top-ranking recent algorithms to tackle the denoising problem.

Keywords: ECG signals, set of wavelets, translation-invariant, Donoho's denoising, noisy copies generation from a single record, white Gaussian noise.

### 1. Introduction

Nowadays, healthcare systems that diagnose and detect human pathologies play a significant role in our modern daily individual lives. Such systems must be efficient and accurate to process and analyse several bio-signals (ECG, EEG, EMG, PCG, *etc.*) in a manner that avoids false analysis and

wrong judgment. Among the problems that can be encountered in the design of health monitoring tools there is the noise or disturbance acting in the data acquisition phase and/or in distant transmissions. Consequently, the ECG signal, which is the electrical activity of the human heart, can be noised by many sources (power line interference, contact noise, patient-electrode motion artifacts, EMG noise, baseline drift, *etc.*) [1]. To correctly decide about the cardiac health status, the noise must be, as much as possible, efficiently reduced. In this context, several contributions have been accomplished. One of the noise types commonly treated is the *additive white Gaussian noise* (AWGN). Essentially, we can categorize previous works relating to the subject as follows:

- Techniques derived from image processing, such as morphological filters. Such methods use erosion, dilation, opening, and closing [2–4].
- Strategies based on ICA and/or PCA, as cited in [5–7].
- Filtering algorithms such as the methods using an adaptive filtering system [8–12].
- Methods using the recent transform known as *Empirical Mode Decomposition* (EMD) [13–15].
- Reported schemes based on the wavelet transform, such as those mentioned in [16, 17].
- Combination-based methodology merging such as Wiener/Kalman filters [18], wavelet/Savitzky–Golay filter [19], wavelet/Wiener filters [20], wavelet/fuzzy reasoning [21].

However, we would like attract the reader’s attention to the algorithms based on the special wavelet-thresholding case, which can be considered as variants of the standard state-of-the-art (Donoho’s) algorithm [22, 23]. In [24], the authors presented a method based on nonlinear thresholding of wavelets and optimized wavelet packet coefficients using hard and soft thresholding, investigating four well-known strategies (Sure, Heuristic Sure, Fixthres, and Minimax). As reported, the application of wavelet denoising methodologies to white noise showed performance superior to that of wavelet packets. Also, in [25], a scheme based on second-generation (Lifting scheme) wavelets with level-dependent thresholds determination was suggested. The authors reported that the obtained results depend on the wavelet type filters, the applied thresholding method, and the decomposition depth. The method has superior performance when faced with the median filter and is faster than traditional wavelet decomposition. Another wavelet-based methodology is given in [26]. It proffers, mainly, a wavelet-based scheme to denoise corrupted ECG records by applying conventional wavelet soft-thresholding (shrinkage) using a level of decomposition 8 (selected empirically) and the Daubechies wavelet Db4 (of 4 vanishing moments). Along the same lines, the contribution shown in [27] consists of recovering a clean ECG signal from its noisy version using a subband-dependent threshold. The so-called S-median thresholds calculated by this technique can be considered as a variant of the well-known Donoho’s universal threshold. Moreover, in [28], the presented contribution is based on the genetic algorithm’s optimization of several parameters that allow the successful cleaning of a noisy ECG. These parameters are, as mentioned, the mother wave, the decomposition level, the type and the rule of the threshold, and finally the rescaling approach. Another denoising strategy, developed by Y. Yang *et al.* [29], consists of reducing white Gaussian noise and particularly suppressing the pseudo-Gibbs phenomena with universal threshold by the *random interpolation average* (RIA) technique. In the latter, independent denoised signals were obtained by applying interpolation and denoising each time. The pseudo-Gibbs phenomenon was suppressed by calculating the mean of all the independent denoised signals. In [30], the signal denoising method was based on translation-invariant, *discrete wavelet transform* (DWT) and *goodness-of-fit* (GOF) tests. This approach performs on several scales; it consists of determining which DWT coefficients represent noise and removing them by using GOF statistical examinations. Additionally, Naveed *et al.* [31] adopted a strategy that is based on the combination of statistical neighbourhood dependencies of DWT coefficients and the GOF test. This allows classifying coefficients as signal or as Gaussian noise. Also, in [32], Talbi made a strategy applied to ECG signals based on the one-dimensional double-density complex DWT noise reduction technique in

the *stationary bionic wavelet transform* (SBWT) domain. Furthermore, Zhang *et al.* [33] presented a method for denoising of ECG signals contaminated with white noise. This method is achieved through combining wavelet energy with a smoothing filter. Moreover, Liu *et al.* [34] proposed an ECG denoising approach based on the *basis pursuit* technique (BP) and *alternating direction method of multipliers* (ADMM) optimization. In this recent work, the combination of low-pass filtering and compressed sensing recovery achieved a high signal-to-noise ratio.

Based on the problem statement and reported literature survey, we present in this work an improved version of the standard state-of-the-art translation-invariant method [35] showing its superior performance compared to some powerful recent contributions. The main idea of the proposed scheme is the use of a set of mother waves in the standard Donoho’s algorithm and it is noted that each mother wave gives a resulting less noisy ECG record. Consequently, it will be shown by the results that averaging the resulting less noisy ECG signals will significantly improve the recovery process.

The paper is organized as follows: Section 2 which presents the standard algorithm of Donoho [22, 23]. In Section 3, the translation-invariant technique is resumed. In Section 4, the proposed improved methodology is described in detail. Results followed by thorough discussions are given in Sections 5 and 6 respectively. Finally, main results and concluding remarks with possible future work are summarized in Section 7.

## 2. Donoho’s standard wavelet-thresholding-based algorithm and other reference denoising methods

The conventional statement of the problem of denoising a noisy signal contaminated with *additive white Gaussian noise* (AWGN) is usually given by:

$$Ns(i) = Fs(i) + n(i)/i = 0, 1, \dots, L - 1, \quad (1)$$

where:  $Ns$  is the noisy signal of length  $L$ ,  $Fs$  is the unknown free-noise deterministic signal of length  $L$ , to be recovered (estimated),  $n$  are the  $L$  samples of an i.i.d. (*independent and identically distributed*) white Gaussian noise that follows the probability density function  $N(0, \sigma^2)$  [22].  $N(0, \sigma^2)$  is the normal distribution of zero mean and standard deviation  $\sigma$ .

The Donoho’s method [22] for denoising can be summarized as follows:

- Application of the *discrete wavelet transform* (DWT) on the noisy signal was introduced by Mallat in [36].
- Thresholding of wavelet coefficients.
- Recovery of the estimated unknown signal by applying the inverse DWT.

The two well-known thresholding strategies that were proposed by Donoho *et al.* [22] are presented by the following:

- Soft thresholding strategy:

$$WCTH(i) = \begin{cases} WC(i) - TH & WC(i) \geq TH. \\ 0 & |WC(i)| < TH. \\ WC(i) + TH & WC(i) \leq -TH. \end{cases} \quad (2)$$

$WC$  is the wavelet coefficients vector of the noisy signal  $Ns$ .  $WCTH$  is the thresholded wavelet coefficient vector resulting from the thresholding process using the threshold  $TH$ .

- Hard thresholding strategy:

$$WCTH(i) = \begin{cases} WC(i) & |WC(i)| \geq TH. \\ 0 & |WC(i)| < TH. \end{cases} \quad (3)$$

Several methods for the choice of the threshold(s) have been suggested in the specialized literature. The well-known standard universal threshold [22] is described by:

$$TH = \sigma\sqrt{2 \log(L)} \tag{4}$$

Also, we can mention other methods based on the minimax estimation principle [23].

For practical use in the real world,  $\sigma$  is actually unknown and should be appropriately estimated. A valid estimation is shown in the following equation [22, 23]:

$$\sigma = \frac{MAD(CD_1)}{0.6745}, \tag{5}$$

where MAD is the abbreviation of *Median Absolute Deviation* and  $CD_1$  are the fine-scale details wavelet coefficients.

In the reported methods, we can see two thresholding rules: the first one is the global or fixed threshold rule and the second one is the level-dependent threshold strategy.

### 3. Overview of the translation-invariant denoising principle

Translation-invariant “TI” denoising is a technique used to address the issue of translation dependency in traditional wavelet-based denoising methods that incorporate the down-sampling (decimation) and the up-sampling operators. In the traditional wavelet transform, the basic functions are not invariant to translations, which can result in artifacts near discontinuities. Translation-invariant denoising methods can better preserve the details and edges of the signal while effectively reducing noise. On the other hand, these methods can be computationally more demanding compared to traditional wavelet-based methods.

Visual artifacts, such as Gibbs phenomena near discontinuities, can indeed be a result of the lack of translation invariance in the wavelet basis. One method to address this and reduce such artifacts is called “cycle spinning”, a term coined by Coifman *et al.* [35]. The idea behind cycle spinning is to “average out” the translation dependence by iteratively applying translations to the signal and computing the average of the transformed signals. By performing this iterative translation and the averaging process, cycle spinning aims to mitigate the translation dependency and reduce visual artifacts in the denoised signal.

The TI technique (Fig. 1) involves applying a range of shift-rotation operations to a signal and then averaging the results to produce a reconstruction with reduced noise phenomena. Averaging the results from different shift-rotation operations allows a more robust and accurate reconstruction.

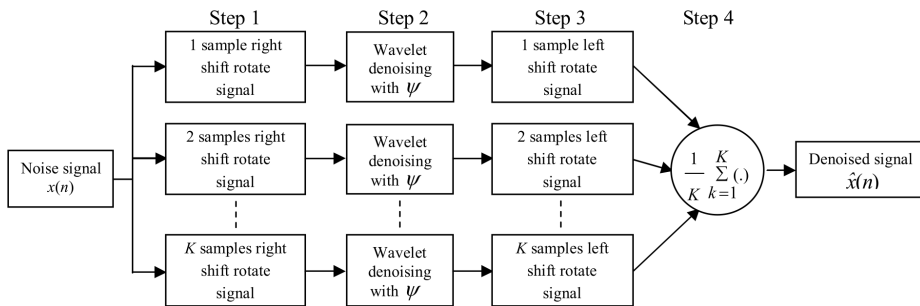


Fig. 1. Principle of the translation-invariant technique.

In Fig. 1, the denoised signal  $\hat{x}(n)$ , involving the use of the conventional translation-invariant strategy, is obtained by using four steps. In Step 1, a range of all circulant right shifts is applied to the noised signal  $x(n)$ . Donoho’s denoising with a fixed one-wavelet mother  $\psi$  is used in Step 2. Next, a range of all circulant right shifts is applied. Finally, an average of the obtained results is calculated to reconstruct  $\hat{x}(n)$ .

#### 4. Proposed low computational method description

The proposed method is essentially based on the work of Coifman *et al.* [35] “translation-invariant” which is well-known as the universal threshold with cycle spinning, in which we use the hard universal thresholding previously mentioned. The main step used in the suggested method is inspired by the well-known noise cancellation problem in the case of multi-sensor data fusion. The rightful question that one can ask is: How can we obtain multiple copies from a single noisy data signal? Equivalently, in other words, how can we reproduce a multi-sensor environment from a real single sensor? The answer is the generation from the original single signal record of different noisy signals by using the standard Donoho’s noise reducer algorithm, and finally, in order to improve noise cancellation, the data-fusion principle is used.

##### 4.1. Multiple copies generation of noisy signals from a single noisy signal

As it is well established, the universal or minimax thresholding algorithm aims to reduce the additive noise. It means that in real cases, the noise is not completely eliminated, but its effect is reduced. The key step of our suggested variant of the translation-invariant standard algorithm is the use of  $K$  several mother waves to generate  $K$  new noisy signals under the assumption that each newly generated noisy record contains a deterministic unknown noise-free signal and a realization of new additive i.i.d. noise produced by the same new process. Note that in the TI algorithm just one mother wave with two ranges of shift rotations is used. With our proposed algorithm a low computational strategy is applied without any shift or rotation.

Let  $\psi_1, \psi_2, \dots, \psi_K$  be  $K$  mother waves. We apply, for each one, the universal or minimax thresholding algorithm to obtain the new resulting noisy signal according to the schematic shown in Fig. 2.

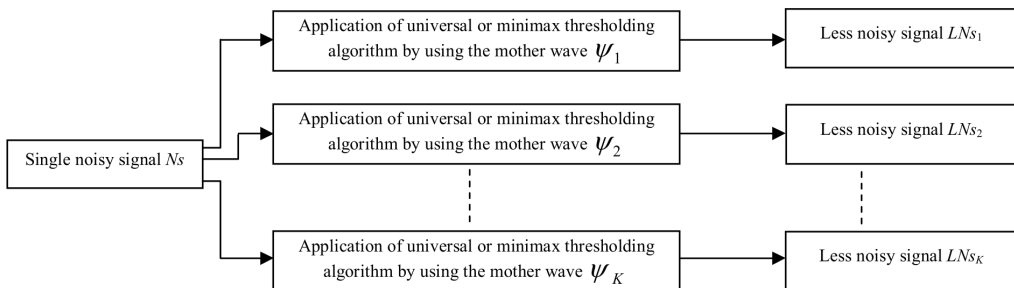


Fig. 2. Multiple-copies Noisy signal generation from an original single-sensored noisy signal.

Each  $k^{\text{th}}$  less noisy signal  $LN_{S_k}/k = 1, 2, \dots, K$ , is modelled following (6):

$$LN_{S_k} = Fs + Nn_k, \tag{6}$$

where:  $LN_{S_k}$  is the  $k^{\text{th}}$  less noisy generated signal, of length  $L$ , by the use of the mother wave  $\psi_k$  in the universal or minimax thresholding algorithm,  $Fs$  is, as mentioned to describe (1),

the unknown free-noise deterministic signal, of length  $L$ , to be recovered (estimated),  $Nn_k$  is the  $k^{\text{th}}$  new realization of a vector noise signal (process  $Nn$ ) of length  $L$ . The new random process  $Nn$  is assumed to be i.i.d. zero mean process. It means that each  $i^{\text{th}}$  sample of the noise process is zero mean ( $E(Nn(i)) = \mu_{Nn}(i) \rightarrow 0$ ).

Compared to translation-invariant, our proposed technique eliminates all the spinning cycles and uses multiwavelet thresholding.

#### 4.2. Improvement of noise cancellation through the data fusion principle

The recovered estimated signal using the averaging technique (sample by sample) approaches the most possible free-noise deterministic unknown signal with respect of the following equation:

$$E(LNs(i)) = E(Fs(i) + E(Nn(i)))/i = 0, 1, \dots, L - 1. \tag{7}$$

The  $i^{\text{th}}$  resulting estimated sample, let it be denoted as  $\hat{F}s(i)$ , is then described by simplifying (7):

$$\hat{F}s(i) = Fs(i) + \mu_{Nn}(i), \tag{8}$$

where:  $\mu_{Nn}(i) \rightarrow 0$  (approaches 0 which is the most possible when using  $K$  mother waves rather than the single best mother wave) leading to the result:

$$\hat{F}s(i) \cong Fs(i). \tag{9}$$

Note that the expectation operator  $E(\cdot)$  can be calculated by:

$$E(\cdot) = \frac{1}{K} \sum_{k=1}^K (\cdot) \tag{10}$$

This well-known technique usually used in the data-fusion discipline [37] can be illustrated in Fig. 3.

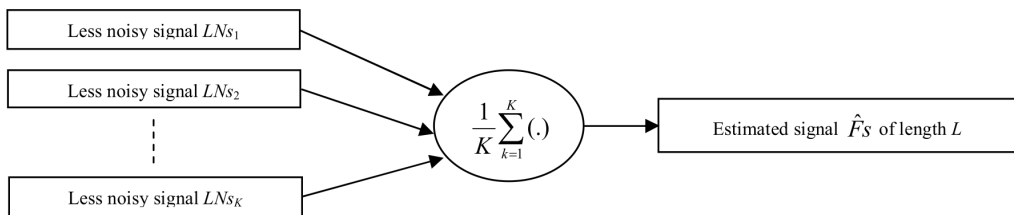


Fig. 3. Noise reduction using the average operator to fuse (sample by sample) the  $K$  less noisy signals ( $LNs_1, LNs_2, \dots, LNs_K$ ).

### 5. Results

The proposed technique was evaluated using the quantitative criteria. The parameters used are SNR\_in (input signal to noise ratio) in dB, SNR\_out (output signal to noise ratio) in dB and MSE (Mean Square Error). They are given respectively by (11), (12) and (13).

$$\text{SNR}_{in} = 10 \log_{10} \left( \frac{\sigma_{ori}^2}{\sigma_{noise}^2} \right), \tag{11}$$

$$\text{SNR}_{\text{out}} = 10 \log_{10} \left( \frac{\sigma_{\text{ori}}^2}{\sigma_{\text{ori-denoised}}^2} \right), \quad (12)$$

where:  $\sigma_{\text{ori}}^2$  represents the variance of the original signal,  $\sigma_{\text{noise}}^2$  is the noise variance,  $\sigma_{\text{ori-denoised}}^2$  denotes the variance of the difference between the clean signal (original) and the denoised one.

$$\text{MSE} = \frac{1}{L} \sum_{n=1}^{N-1} (x(n) - \hat{x}(n))^2 \quad (13)$$

where  $x(n)$  and  $\hat{x}(n)$  are the samples  $n$  of the clean and the denoised signals (of lengths  $L$ ), respectively.

### 5.1. Results relative to synthetic signals

In the first step of our simulations, our approach is evaluated by using four well-known test signals named ‘Blocks’, ‘Bumps’, ‘Heavy Sine’ and ‘Doppler’. We have chosen several lengths ranging between 128 and 8192 samples. The considered corrupting noise is the additive white Gaussian noise with a zero mean and a variance that depends on the SNR\_in. The shapes of noised signals, for SNR\_in equal to 10 dB and with a length of 8192 samples, are shown in Fig. 4.

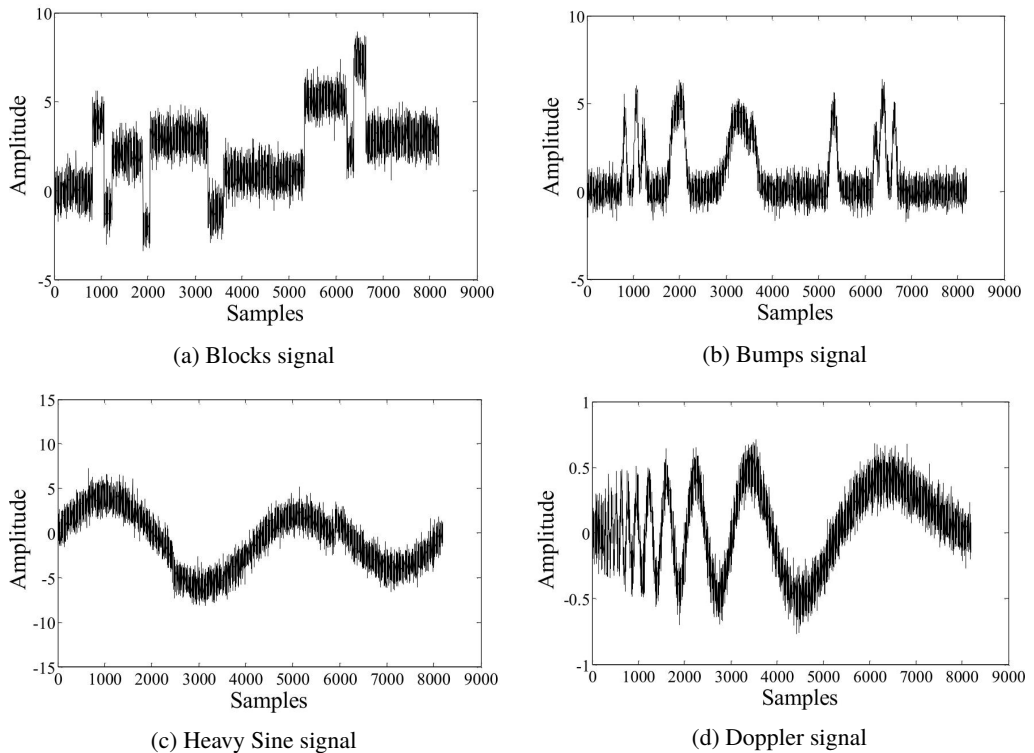


Fig. 4. Shapes of noised signals for signals length = 8192 samples and SNR\_in = 10 dB.

For the denoising process, a set of 21 wavelet mothers {Db1, . . . , Db8, Coif1, . . . , Coif5, Sym1, . . . , Sym8} is used. The employed parameters are the hard universal threshold and wavelet decomposition Level 7.

For the aim of the visual inspection, the waveforms of the denoised signals recovered from the contaminated ones in Fig. 4 are shown in Fig. 5. Therefore, excellent visual quality is noticed for all signals.

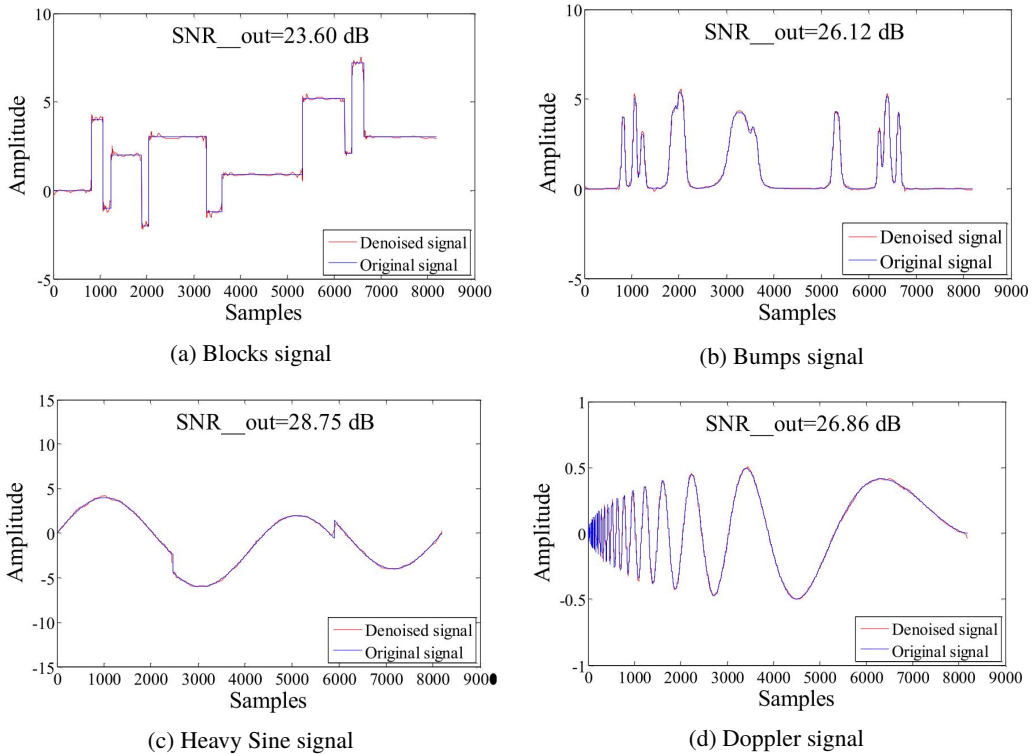


Fig. 5. Shapes of noised signals for signals length = 8192 samples and SNR<sub>in</sub> = 10 dB.

### 5.2. Results relative to ECG signals

Our approach was also applied and evaluated for real ECG signals. These records were obtained from the MIT-BIH Arrhythmia Database [38]. This latter includes a set of 48 types of two-channel ECG recordings, examined by the BIH Arrhythmia Laboratory. The duration of each data is about 30 minutes. All ECG signals were made over a 10 mV range at 360 samples per second with an 11-bit resolution. All types of ECG signals from the Arrhythmia Database were used in our tests.

Note that in this study an additive white Gaussian noise is added to all test signals. Also, for both Donoho’s standard algorithm and the proposed one, the parameters used are the hard minimax threshold and wavelet decomposition Level 4.

Our suggested denoising technique involves 4 sets of wavelets, 3 sets that are: Set 1 {Db1, . . . , Db8}, Set 2 {Coif1, . . . , Coif8}, Set 3 {Sym1, . . . , Sym8} and Set 4 {Db1, . . . , Db8, Coif1, . . . , Coif5, Sym1, . . . , Sym8} constituted from all wavelets belonging to the 3 previously mentioned sets.



In Fig. 6, each  $LN_{sk}$  for  $k = 1, \dots, 21$  (21 wavelet mothers) are generated by using wavelet denoising involving wavelet mothers of Set 4. It means that  $LN_{s1}$  is obtained by employing Db1 wavelet mother,  $LN_{s2}$  is produced by implying Db2, ... and finally,  $LN_{s21}$  is achieved by using Sym8. The estimated sample  $\hat{F}_s(i)$  is the  $i^{\text{th}}$  expectation sample  $E(LN_s(i))/i = 0, 1, \dots, L - 1$ .

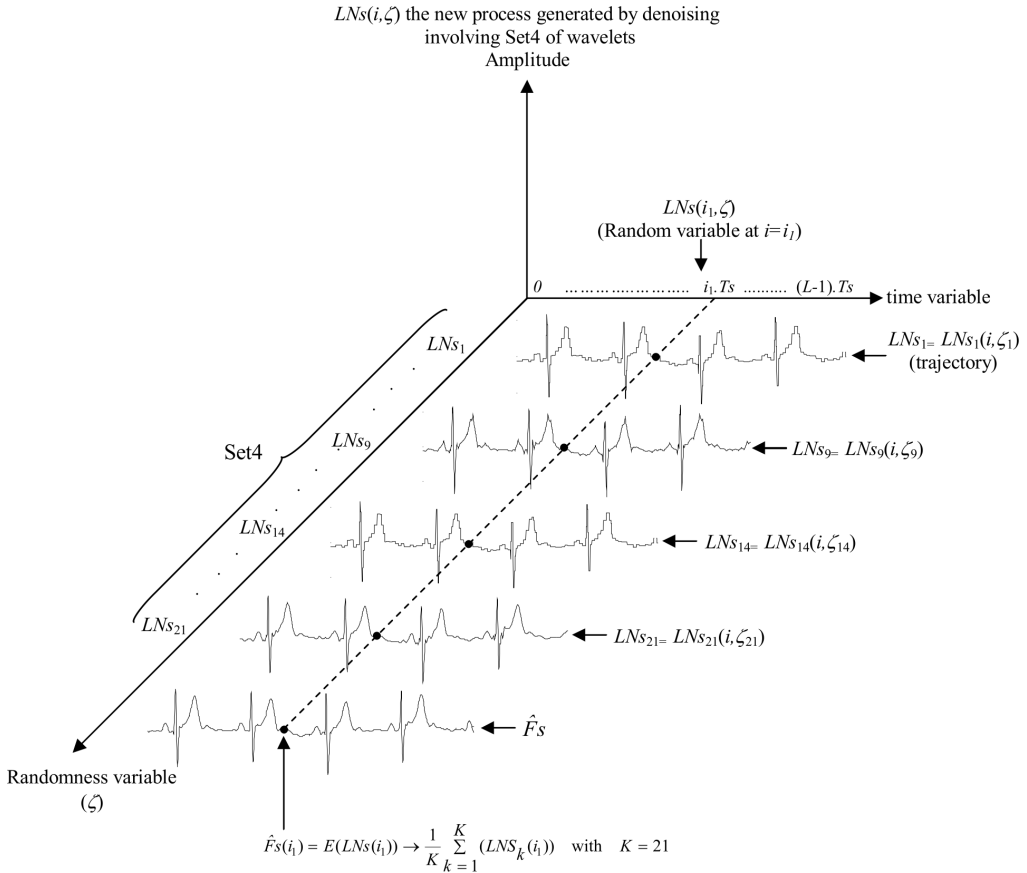


Fig. 6. Illustrative ECG denoising by the suggested strategy implying Set 4.

As it is well known, when a small sample of a signal is used, for example, to confirm a developed denoising strategy, one can repeat the experience many times and take recognizable statistical measures such as the mean and standard deviation because of the variability of the obtained results. However, when the sample (signal) is of a large size, the variability will be significantly reduced to tend to the theoretically probabilistic case. Additionally, as it can be seen that for both methods, the standard deviation is of reduced values ranging from 0.25 dB for a small size of 10s to 0.01 dB for the entire size of the signal (See the illustrative Fig. 7). Also, the standard deviation will be much smaller as the size of the signal is longer. The given remark holds for both TI and the proposed method. Accordingly, our case used signals are of a considerable length of 650000 samples, which gives each executed experience practically the same result.

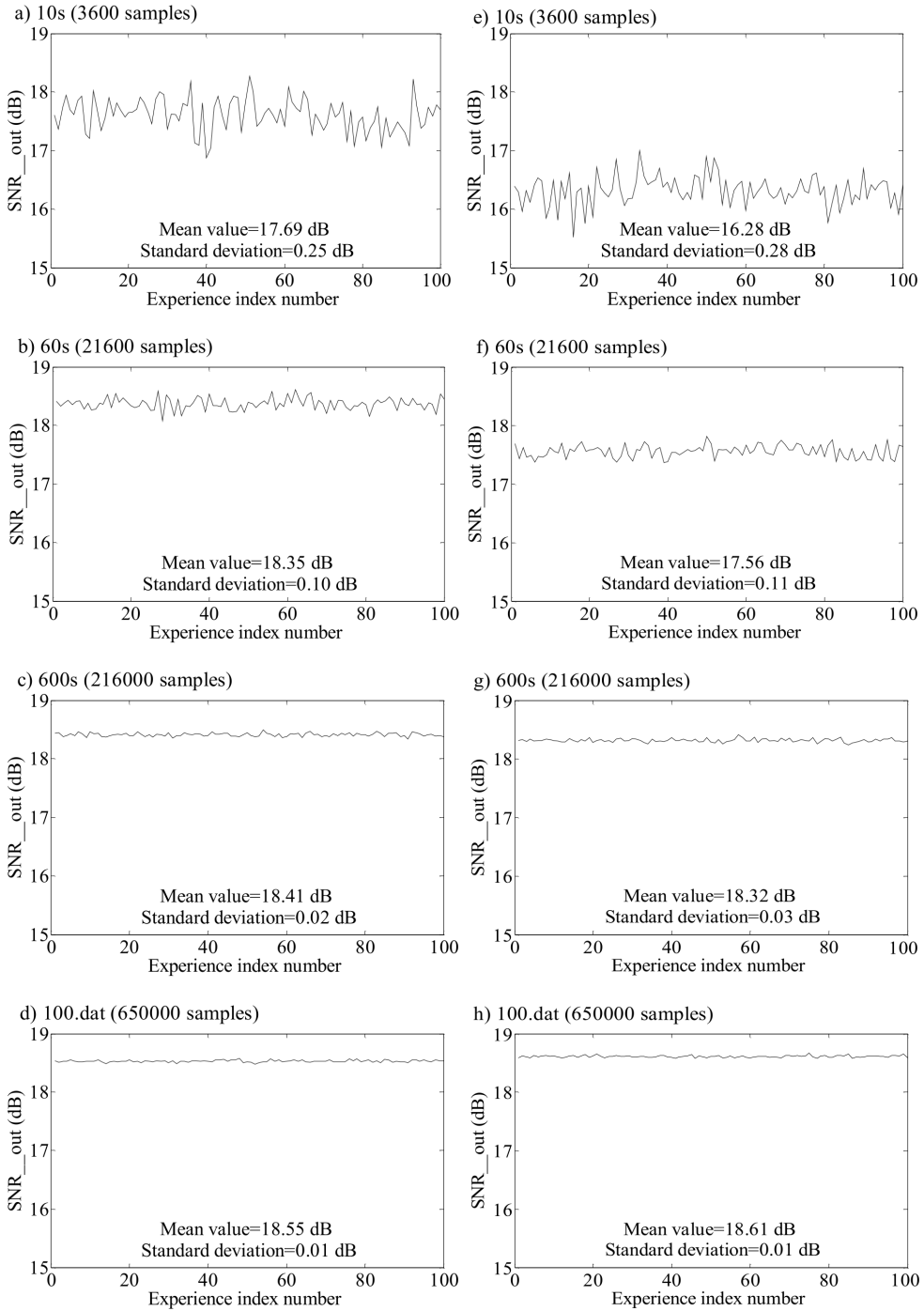


Fig. 7. SNR\_out fluctuations according to the length of the signal 100.dat for SNR\_in = 10 dB.

## 6. Discussion

### 6.1. Discussion related to synthetic signals

Table 1 summarizes the achieved comparison results provided when facing our technique against *translation-invariant* (TI) [35] and RIA (random interpolation average) algorithms [29]. Consequently, one can report that SNR<sub>out</sub> depends on the length of the test signal. Accordingly, in this table, the used lengths for each test signal are 128, 256, 512, 1024, 2048, 4096, and 8192 samples.

Note that the mean value of the SNR<sub>out</sub> of the RIA technique is better than that obtained with TI and our approach for Blocks and Bumps. Thus, the mean values in these cases are equal to 18.57 dB for RIA and 19.39 dB for Blocks and Bumps respectively, in opposition to 17.86 dB and 19.29 dB yielded by our technique. On the other hand, when using our approach, the obtained average values of SNR<sub>out</sub> are 23.40 dB and 20.34 dB for Heavy Sine and Doppler signals respectively, against 23.10 dB and 19.06 dB for TI and 22.57 dB and 19.71 dB for the RIA algorithm. The global mean value for all synthetic signals calculated by our approach is 20.22 dB, versus 19.02 dB and 20.06 dB for TI and RIA, respectively. Therefore, this result shows that our approach outperforms the TI and RIA techniques.

Table 1. SNR<sub>out</sub> of TI, RIA and our approach for SNR<sub>in</sub> of 10 dB for different signal lengths.

Signal length	Blocks			Bumps		
	TI [35]	RIA [29]	Our approach	TI [35]	RIA [29]	Our approach
128	10.97	<b>12.84</b>	12.42	9.35	12.09	<b>13.16</b>
256	11.98	<b>14.72</b>	13.75	10.85	13.15	<b>13.93</b>
512	14.10	<b>16.22</b>	15.52	14.95	<b>17.86</b>	16.36
1024	16.30	<b>18.57</b>	17.95	17.67	19.36	<b>19.43</b>
2048	17.79	<b>20.46</b>	19.97	20.26	<b>22.71</b>	21.93
4096	20.60	<b>22.66</b>	21.87	23.77	<b>24.57</b>	24.12
8192	23.21	<b>24.53</b>	23.60	25.73	26.05	<b>26.12</b>
Mean value of SNR <sub>out</sub>	16.42	<b>18.57</b>	17.86	17.51	<b>19.39</b>	19.29
Signal length	Heavy Sine			Doppler		
	TI [35]	RIA [29]	Our approach	TI [35]	RIA [29]	Our approach
128	16.86	14.89	<b>18.21</b>	10.54	10.71	<b>12.23</b>
256	<b>20.15</b>	18.78	18.42	13.71	14.61	<b>16.03</b>
512	21.22	21.30	<b>22.00</b>	16.85	<b>19.08</b>	18.50
1024	24.19	<b>24.60</b>	24.02	20.36	<b>21.11</b>	20.78
2048	25.40	25.63	<b>25.82</b>	22.95	23.04	<b>23.16</b>
4096	<b>26.82</b>	26.25	26.61	23.46	23.67	<b>24.82</b>
8192	27.10	26.57	<b>28.75</b>	25.58	25.76	<b>26.86</b>
Mean value of SNR <sub>out</sub>	23.10	22.57	<b>23.40</b>	19.06	19.71	<b>20.34</b>

Comparative results were also obtained by using two recently published techniques i.e. a translation-invariant version named TI-DWT-GOF [30] and DTCWT-GOF-NF [31]. Accordingly, the results are given in Table 2. One should also note that the adopted length of each synthetic signal used here is 8192 with SNR<sub>in</sub> = 10 dB. Hence, for the only case of Bumps signal, TI-DWT-GOF has exceeded our approach only by SNR<sub>out</sub> around 1 dB. However, our approach significantly outperforms the cited techniques for Doppler, Heavy Sine, and Blocks signals. Additionally, it can be observed that the average SNR<sub>out</sub> has been improved by about 3 dB better compared to the other techniques.

Table 2. SNR<sub>out</sub> (dB) of denoised signals for input signals length = 8192 and SNR<sub>in</sub> = 10 dB.

Techniques	Blocks	Bumps	Heavy Sine	Doppler	Average
TI-DWT-GOF [30]	19.46	<b>27.27</b>	25.46	22.92	23.77
DTCWT-GOF-NF [31]	20.25	24.93	25.02	23.45	23.41
Proposed Approach	<b>23.60</b>	26.12	<b>28.75</b>	<b>26.86</b>	<b>26.33</b>

### 6.2. Discussion related to ECG signals

Fig. 8a illustrates the shape of a clean 117 ECG signal and Fig. 8b shows its noisy version. However, Fig. 8c represents the restored one by using the Donoho’s single wavelet minimax thresholding strategy involving the mother bior 2.6. However, Fig. 8d exemplifies the shape of the denoised signal by using our approach.

The SNR<sub>in</sub> used in this experiment is fixed to 10 dB. Accordingly, the obtained SNR<sub>out</sub> is 17.06 dB for Donoho’s technique against an SNR<sub>out</sub> of 19.17 dB reached by our strategy when using the set of eight wavelets (Sym1...Sym8). It is clear, as well, by visual inspection, that the restoration process of our approach is better than the conventional Donoho’s.

For a fixed SNR<sub>in</sub> of 10 dB, comparative results are reported in Table 3. The comparison is achieved by facing the Donoho’s standard denoising approach and translation-invariant algorithm to our suggested denoising technique.

Table 3. Corresponding SNR<sub>out</sub> (dB) for 48 MIT-BIH Arrhythmia records using SNR<sub>in</sub> = 10 dB, minimax hard threshold and wavelet decomposition Level 4.

Record	Donoho’s method [22]	TI [35]	Proposed approach for a set of wavelets			
	Single bior 2.6	Single bior 2.6	Set 1	Set 2	Set 3	Set 4
100	16.47	<b>18.61</b>	18.49	17.44	18.36	18.55
101	17.22	<b>19.34</b>	19.10	18.20	19.13	19.25
102	16.08	<b>18.27</b>	17.89	17.05	17.99	18.09
103	17.01	19.48	19.58	18.64	19.60	<b>19.75</b>
104	15.71	<b>18.06</b>	17.49	16.59	17.58	17.68
105	16.94	<b>19.06</b>	18.82	18.39	18.84	18.96
106	18.96	19.11	18.91	18.37	19.04	<b>19.13</b>
107	17.54	<b>19.67</b>	18.96	18.78	19.25	19.25

Continued on next page

Table 3 – Continued from previous page

Record	Donoho's method [22]	TI [35]	Proposed approach for a set of wavelets			
	Single bior 2.6	Single bior 2.6	Set 1	Set 2	Set 3	Set 4
108	<b>19.25</b>	18.29	17.90	17.67	17.96	17.99
109	17.73	19.93	19.74	19.73	19.84	<b>19.99</b>
111	16.73	<b>18.86</b>	18.41	18.18	18.57	18.66
112	16.56	<b>18.76</b>	18.44	17.93	18.58	18.66
113	17.21	19.63	19.55	18.61	19.59	<b>19.71</b>
114	15.80	<b>17.69</b>	17.51	16.79	17.52	17.60
115	17.17	19.61	19.78	18.76	19.79	<b>19.97</b>
116	17.28	19.67	19.73	18.99	19.83	<b>19.98</b>
117	17.06	19.11	19.14	18.68	19.17	<b>19.27</b>
118	15.82	<b>17.92</b>	17.41	16.94	17.64	17.66
119	17.36	19.64	19.81	19.11	19.78	<b>19.94</b>
121	18.16	20.07	20.12	20.06	20.24	<b>20.29</b>
122	16.68	18.97	18.84	18.48	19.08	<b>19.16</b>
123	17.05	19.15	19.37	18.68	19.38	<b>19.52</b>
124	17.87	19.96	19.82	19.95	20.22	<b>20.24</b>
200	16.42	<b>18.43</b>	18.00	17.60	18.11	18.20
201	17.35	19.49	19.43	18.79	19.37	<b>19.55</b>
202	17.70	19.71	19.56	19.11	19.63	<b>19.78</b>
203	16.54	<b>18.61</b>	18.19	18.07	18.33	18.40
205	15.76	18.49	18.50	17.63	18.57	<b>18.72</b>
207	17.70	<b>19.56</b>	19.35	19.14	19.39	19.47
208	16.98	<b>19.25</b>	18.87	18.44	19.04	19.13
209	15.42	<b>17.75</b>	17.47	16.69	17.54	17.68
210	17.05	<b>19.06</b>	18.81	18.41	18.92	19.03
212	15.25	<b>17.51</b>	17.28	16.43	17.24	17.38
213	16.59	<b>19.04</b>	18.77	18.13	18.83	19.00
214	17.73	19.76	19.70	19.23	19.70	<b>19.88</b>
215	15.29	<b>17.59</b>	17.16	16.58	17.28	17.39
217	17.74	<b>19.79</b>	19.37	19.21	19.61	19.67
219	17.72	19.98	20.09	19.38	20.10	<b>20.26</b>
220	16.86	19.18	19.18	18.15	19.28	<b>19.43</b>
221	17.04	19.28	19.29	18.58	19.28	<b>19.45</b>
222	15.67	<b>17.75</b>	17.51	16.78	17.63	17.71
223	17.54	19.89	19.68	19.14	19.87	<b>19.98</b>
228	16.75	<b>18.51</b>	18.10	17.92	18.24	18.27
230	16.75	19.30	19.31	18.49	19.27	<b>19.46</b>
231	16.94	19.23	19.41	18.43	19.32	<b>19.51</b>
232	15.58	17.49	<b>17.61</b>	16.93	17.37	17.58
233	17.16	19.47	19.24	19.01	19.40	<b>19.54</b>
234	16.83	19.19	19.29	18.68	19.41	<b>19.56</b>
<b>Average SNR_out</b>	16.70	19.00	18.83	18.27	18.91	<b>19.02</b>

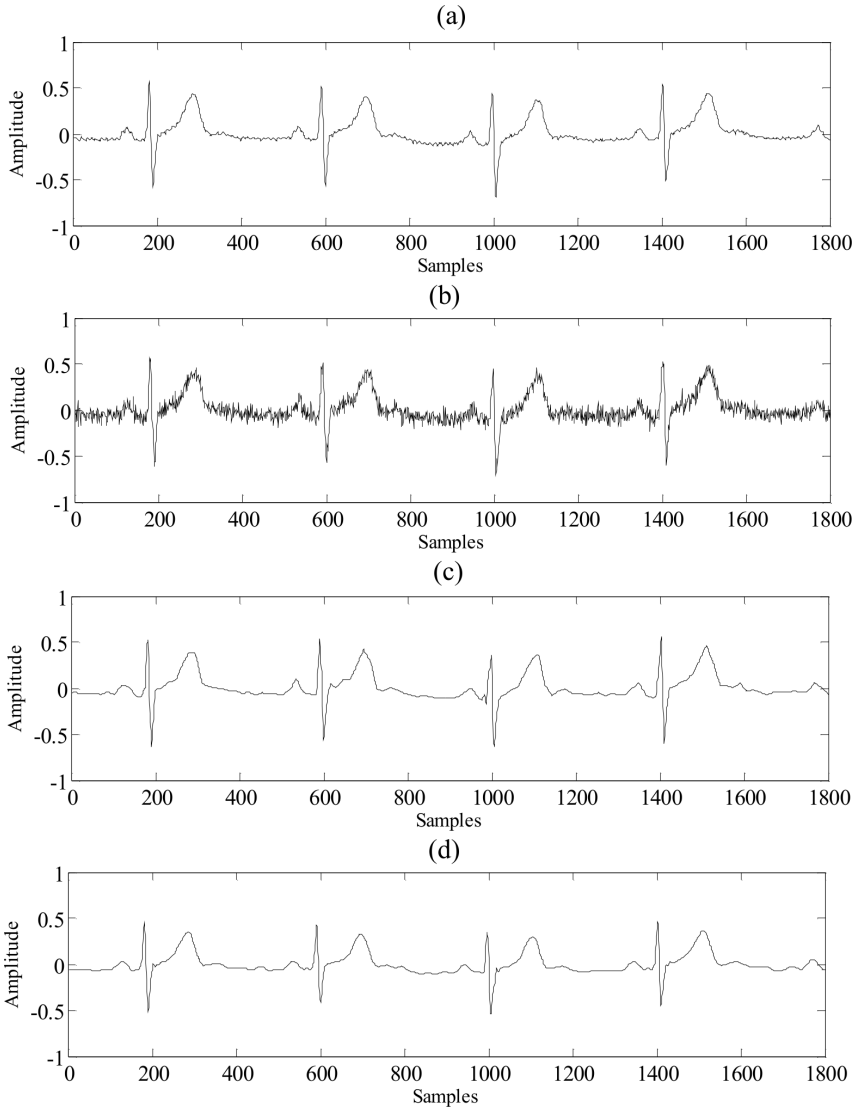


Fig. 8. Noise cancellation of 117 ECG signal, (a) Original signal, (b) Noised signal with SNR<sub>in</sub> = 10 dB, (c) Denoising signal using Donoho's algorithm [22] (SNR<sub>out</sub> = 17.06 dB), (d) Denoising signal using our approach (SNR<sub>out</sub> = 19.17 dB).

Accordingly, Table 3 shows that the average SNR<sub>out</sub> of all ECG signals is 16.70 dB when using the conventional Donoho's algorithm against average SNR<sub>out</sub> of 18.83 dB, 18.27 dB and 18.91 dB when using the proposed technique involving {Db1, ..., Db8}, {Coif1, ..., Coif5} and {Sym1, ..., Sym8} respectively. It is noticeable that the set of wavelets {Sym1, ..., Sym8} offers the best result and reports a significant improvement of 2.21 dB compared to the Donoho's standard algorithm involving single wavelet bior 2.6. When using Set 4, our approach improves the average SNR<sub>in</sub> by 9.02 dB (SNR<sub>out</sub> is equal to 19.02 dB).

The mean value of SNR<sub>out</sub> using the TI approach is equal to 19.00 dB, compared to 19.02 dB when applying our algorithm using Set 4. Both methods give practically the same results when the taken length is considerably large.

Additionally, it is worth noting that for shorter lengths of signals, for example when using all the 48 records with a length = 10 s and with SNR<sub>in</sub> = 10 dB, we remarked that our strategy reached an average SNR<sub>out</sub> of 17.57 dB against 16.39 dB related to the TI algorithm.

In terms of complexity, to face the computational efficiency of our approach against one of the TI and RIA algorithms, we can draw up Table 4. The number of operations for  $K$  soft-copies generation needed by our technique is  $K$  wavelet denoising operations and an average value calculation operation. Conversely, TI needs  $2K$  shift rotations in addition to the operations that our approach requests. For the RIA technique, we need  $K$  wavelet denoising operations,  $2K$  operations of interpolations and an operation for calculating the mean value. We can state that our approach is the lowest in terms of computational complexity for practically same quality.

Table 4. Number of operations of  $K$  soft copies generation.

Operations	Wavelet denoising	Right shift rotation	Left shift rotation	Interpolation	Inverse interpolation	Mean value operation
Proposed method	$K$	–	–	–	–	1
TI [35]	$K$	$K$	$K$	–	–	1
RIA [29]	$K$	–	–	$K$	$K$	1

We have also assessed and contrasted our method with a recent technique proposed by M. Talbi in 2020 [32] that is regarded as one of the best methods for denoising ECG signals corrupted by white Gaussian noise. The algorithm involves the use of the *stationary bionic wavelet transform* (SBWT) range to apply one-dimensional double-density complex DWT denoising. Simulation results (of seven noised signals 100.dat to 106.dat) in Table 5 are provided from the calculations of SNR<sub>out</sub> values and their corresponding MSE for SNR<sub>in</sub> range from –5 dB and 15 dB. Each SNR<sub>out</sub> represents the average value of seven values of SNR<sub>out</sub> related to the cited signals and each MSE is also the mean value of seven MSE computed by using equation (13). The outcomes prove that for all values used of SNR<sub>in</sub>, our method performs better than Talbi’s algorithm.

Table 5. Results comparison between our approach and Talbi’s algorithm.

SNR <sub>in</sub> (dB)	-5	0	5	10	15
SNR <sub>out</sub> (dB) of our approach	<b>5.65</b>	<b>10.15</b>	<b>14.72</b>	<b>18.77</b>	<b>22.24</b>
SNR <sub>out</sub> (dB) of Talbi [32]	5.25	9.71	14.08	18.09	21.66
MSE of our approach	<b>3.97e-04</b>	<b>1.42e-04</b>	<b>4.96e-05</b>	<b>1.94e-05</b>	<b>8.74e-06</b>
MSE of Talbi [32]	0.0071	0.0026	9.4286e-04	3.7143e-04	1.5714e-04

In Table 6, we also showed a comparison of our method using Set 4 with two techniques based on translation-invariant: Zhang *et al.* [33] and the most recent published paper of Liu *et al.* [34].

Table 6. SNR\_out (dB) comparison results of SNR\_in = 5 dB.

Records	Zhang <i>et al.</i> [33]	Liu <i>et al.</i> [34]	Proposed Method
100	12.10	–	14.39
103	12.90	–	15.16
105	<b>14.40</b>	11.32	14.29
106	–	10.89	<b>14.61</b>
107	–	<b>16.67</b>	14.29
108	–	9.89	<b>14.46</b>
109	–	13.16	<b>15.03</b>
111	–	8.67	<b>13.31</b>
112	–	<b>15.01</b>	13.97
113	12.70	11.70	<b>15.11</b>
114	–	8.51	<b>13.21</b>
115	13.00	12.90	<b>15.22</b>
116	–	<b>18.65</b>	14.53
117	13.00	<b>16.81</b>	14.90
118	–	<b>18.16</b>	13.14
119	12.80	<b>17.91</b>	14.60
122	13.50	–	14.44
200	12.70	11.70	<b>13.80</b>
201	–	9.02	<b>14.86</b>
202	–	8.75	<b>15.04</b>
203	–	11.92	<b>13.96</b>
205	–	11.08	<b>14.30</b>
207	–	10.63	<b>15.15</b>
208	–	13.15	<b>13.86</b>
209	–	9.76	<b>13.55</b>
210	–	9.13	<b>14.50</b>
212	–	10.12	<b>13.51</b>
215	<b>13.60</b>	–	12.99
230	12.00	–	<b>14.63</b>
231	–	9.56	<b>15.06</b>
232	–	8.65	<b>14.10</b>
233	–	14.21	<b>14.50</b>
234	–	9.55	<b>14.28</b>
<b>Average SNR_out</b>	12.97	12.05	<b>14.32</b>

Knowing that the SNR\_in is 5 dB, the average SNR\_out obtained by Zhang *et al.* is equal to 12.97 dB which is greater than obtained one by Liu *et al.* (12.05 dB). However, our technique offers better results than these two approaches, with SNR\_out equal to 14.32 dB.



## 7. Conclusions

In this paper, a new translation-invariant-like strategy based on involving a set of wavelets in the denoising process is proposed. The strong point of our work is the generation of noisy soft copies from only one acquired noisy signal, imitating the multisensor acquisition scenario. Additionally, it is less complicated than the TI [35], strategy which uses an additional cycle spinning operation for each soft copy generation and the RIA [29] technique which implies additional interpolation for each soft copy generation. Finally, our method, when compared to the different algorithms reported in [22, 29–35] shows performance superior than the other techniques. Thus, the suggested approach can be considered as an effective, low-cost solution avoiding the use of multisensors to improve noise cancellation. Finally, as a future research direction, we will aim at reducing powerline interference and baseline wander noises by associating a technique dedicated to such types of noise with our strategy.

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