

Temporal noise shaping

Description

5 1. Technical field

Examples herein relate to encoding and decoding apparatus, in particular for performing temporal noise shaping (TNS).

2. Prior art

The following prior art documents are in the prior art:

- 10 [1] Herre, Jürgen, and James D. Johnston. "Enhancing the performance of perceptual audio coders by using temporal noise shaping (TNS)." Audio Engineering Society Convention 101. Audio Engineering Society, 1996.
- [2] Herre, Jürgen, and James D. Johnston. "Continuously signal-adaptive filterbank for high-quality perceptual audio coding." Applications of Signal Processing to
15 Audio and Acoustics, 1997. 1997 IEEE ASSP Workshop on. IEEE, 1997.
- [3] Herre, Jürgen. "Temporal noise shaping, quantization and coding methods in perceptual audio coding: A tutorial introduction." Audio Engineering Society Conference: 17th International Conference: High-Quality Audio Coding. Audio Engineering Society, 1999.
- 20 [4] Herre, Jürgen Heinrich. "Perceptual noise shaping in the time domain via LPC prediction in the frequency domain." U.S. Patent No. 5,781,888. 14 Jul. 1998.
- [5] Herre, Jürgen Heinrich. "Enhanced joint stereo coding method using temporal envelope shaping." U.S. Patent No. 5,812,971. 22 Sep. 1998.
- [6] 3GPP TS 26.403; General audio codec audio processing functions; Enhanced
25 aacPlus general audio codec; Encoder specification; Advanced Audio Coding (AAC) part.

[7] ISO/IEC 14496-3:2001; Information technology – Coding of audio-visual objects – Part 3: Audio.

[8] 3GPP TS 26.445; Codec for Enhanced Voice Services (EVS); Detailed algorithmic description.

- 5 Temporal Noise Shaping (TNS) is a tool for transform-based audio coders that was developed in the 90s (conference papers [1-3] and patents [4-5]). Since then, it has been integrated in major audio coding standards such as MPEG-2 AAC, MPEG-4 AAC, 3GPP E-AAC-Plus, MPEG-D USAC, 3GPP EVS, MPEG-H 3D Audio.
- 10 TNS can be briefly described as follows. At the encoder-side and before quantization, a signal is filtered in the frequency domain (FD) using linear prediction, LP, in order to flatten the signal in the time-domain. At the decoder-side and after inverse quantization, the signal is filtered back in the frequency-domain using the inverse prediction filter, in order to shape the quantization noise in the
- 15 time-domain such that it is masked by the signal.

TNS is effective at reducing the so-called pre-echo artefact on signals containing sharp attacks such as e.g. castanets. It is also helpful for signals containing pseudo stationary series of impulse-like signals such as e.g. speech.

- 20 TNS is generally used in an audio coder operating at relatively high bitrate. When used in an audio coder operating at low bitrate, TNS can sometimes introduce artefacts, degrading the quality of the audio coder. These artefacts are click-like or noise-like and appear in most of the cases with speech signals or tonal music signals.

- 25 Examples in the present document permit to suppress or reduce the impairments of TNS maintaining its advantages.

Several examples below permit to obtain an improved TNS for low-bitrate audio coding.

3. Summary of the invention

In accordance with examples, there is provided an encoder apparatus comprising:

a temporal noise shaping, TNS, tool for performing linear prediction, LP, filtering on an information signal including a plurality of frames; and

5 a controller configured to control the TNS tool so that the TNS tool performs LP filtering with:

a first filter whose impulse response has a higher energy; and

a second filter whose impulse response has a lower energy than the impulse response of the first filter, wherein the second filter is not an identity
10 filter,

wherein the controller is configured to choose between filtering with the first filter and filtering with the second filter on the basis of a frame metrics.

It has been noted that it is possible to remove artefacts on problematic frames while minimally affecting the other frames.

15 Instead of simply turning on/off the TNS operations, it is possible to maintain the advantages of the TNS tool while reducing its impairments. Therefore, an intelligent real-time feedback-based control is therefore obtained by simply reducing filtering where necessary instead of avoiding it.

In accordance with examples, the controller is further configured to:

20 modify the first filter so as to obtain the second filter in which the filter's impulse response energy is reduced.

Accordingly, the second filter with reduced impulse response energy may be created when necessary.

In accordance with examples, the controller is further configured to:

apply at least one adjustment factor to the first filter to obtain the second filter.

- By intelligently modifying the first filter, a filtering status may be created which is not be achievable by simply performing operations of turning on/off the TNS. At
 5 least one intermediate status between full filtering and no filtering is obtained. This intermediate status, if invoked when necessary, permits to reduce the disadvantages of the TNS maintaining its positive characteristics.

In accordance with examples, the controller is further configured to:

- define the at least one adjustment factor on the basis of at least the frame
 10 metrics.

In accordance with examples, the controller is further configured to:

define the at least one adjustment factor on the basis of a TNS filtering determination threshold which is used for selecting between performing TNS filtering and non-performing TNS filtering.

- 15 In accordance with examples, the controller is further configured to:

define the at least one adjustment factor using a linear function of the frame metrics, the linear function being such that an increase in the frame metrics corresponds to an increase of the adjustment factor and/or of the filter's impulse response energy.

- 20 Therefore, it is possible to define, for different metrics, different adjustment factors to obtain the filter parameters which are the most appropriated for each frame.

In accordance with examples, the controller is further configured to define the adjustment factor as

$$\gamma = \begin{cases} 1 - (1 - \gamma_{\min}) \frac{\text{thresh2} - \text{frameMetrics}}{\text{thresh2} - \text{thresh}} & , \text{ if frameMetrics} < \text{thresh2} \\ 1 & , \text{ otherwise} \end{cases}$$

wherein thresh is the TNS filtering determination threshold, thresh2 is the filtering type determination threshold, frameMetrics is a frame metrics, and $\gamma_{\min}$ is a fixed value.

Artefacts caused by the TNS occur in frames in which the prediction gain is in a particular interval, which is here defined as the set of values higher than the TNS filtering determination threshold thresh but lower than the filtering determination threshold thresh2. In some cases in which the metrics is the prediction gain, thresh = 1.5 and thresh2 = 2, artefacts caused by the TNS tend to occur between 1.5 and 2. Therefore, several examples permit to overcome these impairments by reducing the filtering for $1.5 < \text{predGain} < 2$.

In accordance with examples, the controller is further configured to modify the parameters of the first filter to obtain the parameters of the second filter by applying:

$$a_w(k) = \gamma^k a(k), k = 0, \dots, K$$

where $a(k)$ are parameters of the first filter, γ is the adjustment factor such that $0 < \gamma < 1$, $a_w(k)$ are the parameters of the second filter and K is the order of the first filter.

This is an easy but valid technique for obtaining the parameters of the second filter so that the impulse response energy is reduced in respect to the impulse response energy of the first filter.

In accordance with examples, the controller is further configured to obtain the frame metrics from at least one of a prediction gain, an energy of the information signal and/or a prediction error.

That these metrics permit to easily and reliably discriminate the frames which need to be filtered by the second filter from the frames which need to be filtered by the first filter.

In accordance with examples, the frame metrics comprises a prediction gain calculated as

$$predGain = \frac{energy}{predError}$$

where *energy* is a term associated to an energy of the information signal, and *predError* is a term associated to a prediction error.

In accordance with examples, the controller is configured so that:

- 5 at least for a reduction of a prediction gain and/or a reduction of an energy of the information signal, the second filter's impulse response energy is reduced, and/or at least for an increase of the prediction error, the second filter's impulse response energy is reduced.

In accordance with examples, the controller is configured to:

- 10 compare the frame metrics with a filtering type determination threshold (e.g., thresh2), so as to perform a filtering with the first filter when the frame metrics is lower than the filtering type determination threshold.

Accordingly, it is easy to automatically establish whether the signal is to be filtered using the first filter or using the second filter.

- 15 In accordance with examples, the controller is configured to:

 choose between performing a filtering and non-performing filtering on the basis of the frame metrics.

Accordingly, it is also possible to completely avoid TNS filtering at all when not appropriated.

- 20 In examples, the same metrics may be used twice (by performing comparisons with two different thresholds): both for deciding between the first filter and second filter, and for deciding whether to filter or not to filter.

In accordance with examples, the controller is configured to:

compare the frame metrics with a TNS filtering determination threshold, so as to choose to avoid TNS filtering when the frame metrics is lower than the TNS filtering determination threshold.

In accordance with examples, the apparatus may further comprise:

- 5 a bitstream writer to prepare a bitstream with reflection coefficients, or a quantized version thereof, obtained by the TNS.

These data may be stored and/or transmitted, for example, to a decoder.

In accordance with examples, there is provided a system comprising an encoder side and a decoder side, wherein the encoder side comprises an encoder
10 apparatus as above and/or below.

In accordance with examples, there is provided a method for performing temporal noise shaping, TNS, filtering on an information signal including a plurality of frames, the method comprising:

- 15 - for each frame, choosing, on the basis of a frame metrics, between filtering with a first filter whose impulse response has a higher energy and filtering with a second filter whose impulse response has an energy lower than the energy of the impulse response of the first filter (14a), wherein the second filter is not an identity filter;
- 20 - filtering the frame using the filtering with the chosen between the first filter and the second filter.

In accordance with examples, there is provided a non-transitory storage device storing instructions which, when executed by a processor, cause the processor to perform at least some of the steps of the methods above and/or below and/or to implement a system as above or below and/or an apparatus as above and/or
25 below.

4. Description of the drawings

Fig. 1 shows an encoder apparatus according to an example.

Fig. 2 shows a decoder apparatus according to an example.

Fig. 3 shows a method according to an example.

Fig. 3A shows a technique according to an example.

Figs 3B and 3C show methods according to examples.

5 Fig. 4 shows methods according to examples.

Fig. 5 shows an encoder apparatus according to an example.

Fig. 6 shows an decoder apparatus according to an example.

Figs. 7 and 8 show encoder apparatus according to examples.

Figs. 8(1)-8(3) show signal evolutions according to examples.

10 5. Examples

Fig. 1 shows an encoder apparatus 10. The encoder apparatus 10 may be for processing (and transmitting and/or storing) information signals, such as audio signals. An information signal may be divided into a temporal succession of frames. Each frame may be represented, for example, in the frequency domain,
15 FD. The FD representation may be a succession of bins, each at a specific frequency. The FD representation may be a frequency spectrum.

The encoder apparatus 10 may, inter alia, comprise a temporal noise shaping, TNS, tool 11 for performing TNS filtering on an FD information signal 13 ($X_s(n)$). The encoder apparatus 10 may, inter alia, comprise a TNS controller 12. The TNS
20 controller 12 may be configured to control the TNS tool 11 so that the TNS tool 11 performs filtering (e.g., for some frames) using at least one higher impulse response energy linear prediction (LP) filtering and (e.g., for some other frames) using at least one higher impulse response energy LP filtering. The TNS controller 12 is configured to perform a selection between higher impulse response energy
25 LP filtering and lower impulse response energy LP filtering on the basis of a metrics associated to the frame (frame metrics). The energy of the impulse

response of the first filter is higher than the energy of the impulse response of the second filter.

The FD information signal 13 ($X_s(n)$) may be, for example, obtained from a modified discrete cosine transform, MDCT, tool (or modified discrete sine transform MDST, for example) which has transformed a representation of a frame
5 from a time domain, TD, to the frequency domain, FD.

The TNS tool 11 may process signals, for example, using a group of linear prediction (LP) filter parameters 14 ($a(k)$), which may be parameters of a first filter 14a. The TNS tool 11 may also comprise parameters 14' ($a_w(k)$) which may be
10 parameters of a second filter 15a (the second filter 15a may have an impulse response with lower energy as compared to the impulse response of the first filter 14a). The parameters 14' may be understood as a weighted version of the parameters 14, and the second filter 15a may be understood as being derived from the first filter 14a. Parameters may comprise, inter alia, one or more of the
15 following parameters (or the quantized version thereof): LP coding, LPC, coefficients, reflection coefficients, RCs, coefficients $rc_i(k)$ or quantized versions thereof $rc_q(k)$, arcsine reflection coefficients, ASRCs, log-area ratios, LARs, line spectral pairs, LSPs, and/or line spectral frequencies, LS, or other kinds of such parameters. In examples, it is possible to use any representation of filter
20 coefficients.

The output of the TNS tool 11 may be a filtered version 15 ($X_f(n)$) of the FD information signal 13 ($X_s(n)$).

Another output of the TNS tool 11 may be a group of output parameters 16, such as reflection coefficients $rc_i(k)$ (or quantized versions thereof $rc_q(k)$).

25 Downstream to the components 11 and 12, a bitstream coder may encode the outputs 15 and 16 into a bitstream which may be transmitted (e.g., wirelessly, e.g., using a protocol such as Bluetooth) and/or stored (e.g., in a mass memory storage unit).

TNS filtering provides reflection coefficients which are in general different from zero. TNS filtering provides an output which is in general different from the input.

Fig. 2 shows a decoder apparatus 20 which may make use of the output (or a processed version thereof) of the TNS tool 11. The decoder apparatus 20 may
5 comprise, inter alia, a TNS decoder 21 and a TNS decoder controller 22. The components 21 and 22 may cooperate to obtain a synthesis output 23 ($\widehat{X}_s(n)$). The TNS decoder 21 may be, for example, input with a decoded representation 25 (or a processed version thereof ($\widehat{X}_f(n)$)) of the information signal as obtained by the decoder apparatus 20. The TNS decoder 21 may obtain in input (as input 26)
10 reflection coefficients $rc_i(k)$ (or quantized versions thereof $rc_q(k)$). The reflection coefficients $rc_i(k)$ or $rc_q(k)$ may be the decoded version of the reflection coefficients $rc_i(k)$ or $rc_q(k)$ provided at output 16 by the encoder apparatus 10.

As shown in Fig. 1, the TNS controller 12 may control the TNS tool 11 on the basis, inter alia, of a frame metrics 17 (e.g., prediction gain or predGain). For
15 example, the TNS controller 12 may perform filtering by choosing between at least a higher impulse response energy LP filtering and/or a lower impulse response energy LP filtering, and/or between filtering and non-filtering. Apart from the higher impulse response energy LP filtering and the lower impulse response energy LP filtering, at least one intermediate impulse response energy LP filtering are
20 possible according to examples.

Reference numeral 17' in Fig. 1 refers to information, commands and/or control data which are provided to the TNS tool 14 from the TNS controller 12. For example, a decision based on the metrics 17 (e.g., "use the first filter" or "use the second filter") may be provided to the TNS tool 14. Settings on the filters may also
25 be provided to the TNS tool 14. For example, an adjustment factor (γ) may be provided to the TNS filter so as to modify the first filter 14a to obtain the second filter 15a.

The metrics 17 may be, for example, a metrics associated to the energy of the signal in the frame (for example, the metrics may be such that the higher the
30 energy, the higher the metrics). The metrics may be, for example, a metrics

- associated to a prediction error (for example, the metrics may be such that the higher the prediction error, the lower the metric). The metrics may be, for example, a value associated to the relationship between the prediction error and energy of the signal (for example, the metrics may be such that the higher the ratio between the energy and the prediction error, the higher the metrics). The metrics may be, for example, a prediction gain for a current frame, or a value associated or proportional to the prediction gain for the current frame (such as, for example, the higher the prediction gain, the higher the metrics). The frame metrics (17) may be associated to the flatness of the signal's temporal envelope.
- 10 It has been noted that artefacts due to TNS occur only (or at least prevalently) when the prediction gain is low. Therefore, when the prediction gain is high, the problems caused by TNS do not arise (or are less prone to arise) and it is possible to perform full TNS (e.g., higher impulse response energy LP). When the prediction gain is very low, it is preferable not to perform TNS at all (non-filtering).
- 15 When the prediction gain is intermediate, it is preferable to reduce the effects of the TNS by using a lower impulse response energy linear prediction filtering (e.g., by weighting LP coefficients or other filtering parameters and/or reflection coefficients and/or using a filter whose impulse response has a lower energy). The higher impulse response energy LP filtering and the lower impulse response
- 20 energy LP filtering are different from each other in that the higher impulse response energy LP filtering is defined so as to cause a higher impulse response energy than the lower impulse response energy LP filtering. A filter is in general characterized by the impulse response energy and, therefore, it is possible to identify it with its impulse response energy. The higher impulse response energy
- 25 LP filtering means using a filter whose impulse response has a higher energy than the filter used in the lower impulse response energy LP filtering.

Hence, with the present examples, the TNS operations may be computed by:

- performing high impulse response energy LP filtering when the metrics (e.g. prediction gain) is high (e.g., over a filtering type determination threshold);

- performing low impulse response energy LP filtering when the metrics (e.g. prediction gain) is intermediate (e.g., between a TNS filtering determination threshold and the filtering type determination threshold); and
- non-performing TNS filtering when the metrics (e.g. prediction gain) is low (e.g., under the TNS filtering determination threshold).

High impulse response energy LP filtering may be obtained, for example, using a first filter having a high impulse response energy. Low impulse response energy LP filtering may be obtained, for example, using a second filter having a lower impulse response energy. The first and second filter may be linear time-invariant (LTI) filters.

In examples, the first filter may be described using the filter parameters $a(k)$ (14). In examples, the second filter may be a modified version of the first filter (e.g., as obtained by the TNS controller 12). The second filter (lower impulse response energy filter) may be obtained by downscaling the filter parameters of the first filter (e.g., using a parameter γ or γ^k such that $0 < \gamma < 1$, with k being a natural number such that $k \leq K$, K being the order of the first filter).

Therefore, in examples, when the filter parameters are obtained, and on the basis of the metrics, it is determined that the lower impulse response energy filtering is necessary, the filter parameters of the first filter may be modified (e.g., downscaled) to obtain filter parameters of the second filter, to be used for the lower impulse selection energy filter.

Fig. 3 shows a method 30 which may be implemented at the encoder apparatus 10.

At step S31, a frame metrics (e.g., prediction gain 17) is obtained.

At step S32, it is checked whether the frame metrics 17 is higher than a TNS filtering determination threshold or first threshold (which may be 1.5, in some examples). An example of metrics may be a prediction gain.

If at S32 it is verified that the frame metrics 17 is lower than the first threshold (thresh), no filtering operation is performed at S33 (it could be possible to say that an identity filter is used, the identity filter being a filter in which the output is the same of the input). For example, $X_r(n) = X_s(n)$ (the output 15 of the TNS tool 11 is the same as the input 13), and/or the reflection coefficients $rc_i(k)$ (and/or their quantized versions $rc_0(k)$) are also set at 0. Therefore, the operations (and the output) of the decoder apparatus 20 will not be influenced by the TNS tool 11. Hence, at S33, neither the first filter nor the second filter may be used.

If at S32 it is verified that the frame metrics 17 is greater than the TNS filtering determination threshold or first threshold (thresh), a second check may be performed at step S34 by comparing the frame metrics with a filtering type determination threshold or second threshold (thresh2, which may be greater than the first threshold, and be, for example, 2).

If at S34 it is verified that the frame metrics 17 is lower than the filtering type determination threshold or second threshold (thresh2), lower impulse response energy LP filtering is performed at S35 (e.g., a second filter with lower impulse response energy is used, the second filter non-being an identity filter).

If at S34 it is verified that the frame metrics 17 is greater than the filtering type determination threshold or second threshold (thresh2), higher impulse response energy LP filtering is performed at S36 (e.g., a first filter whose response energy is higher than the lower energy filter is used).

The method 30 may be reiterated for a subsequent frame.

In examples, the lower impulse response energy LP filtering (S35) may differ from the higher impulse response energy LP filtering (S36) in that the filter parameters 14 ($a(k)$) may be weighted, for example, by different values (e.g., the higher impulse response energy LP filtering may be based on unitary weights and the lower impulse response energy LP filtering may be based on weights lower than 1). In examples, the lower impulse response energy LP filtering may differ from the higher impulse response energy LP filtering in that the reflection coefficients 16 obtained by performing lower impulse response energy LP filtering may cause a

higher reduction of the impulse response energy than the reduction caused by the reflection coefficients obtained by performing higher impulse response energy LP filtering.

Hence, while performing higher impulse response energy filtering at the step S36,
 5 the first filter is used on the basis of the filter parameters 14 (a(k)) (which are therefore the first filter parameters). While performing lower impulse response energy filtering at the step S35, the second filter is used. The second filter may be obtained by modifying the parameters of the first filter (e.g., by weighting with weight less than 1).

10 The sequence of steps S31-S32-S34 may be different in other examples: for example, S34 may precede S32. One of the steps S32 and/or S34 may be optional in some examples.

In examples, at least one of the first and/or second thresholds may be fixed (e.g., stored in a memory element).

15 In examples, the lower impulse response energy filtering may be obtained by reducing the impulse response of the filter by adjusting the LP filter parameters (e.g., LPC coefficients or other filtering parameters) and/or the reflection coefficients, or an intermediate value used to obtain the reflection coefficients. For example, coefficients less than 1 (weights) may be applied to the LP filter
 20 parameters (e.g., LPC coefficients or other filtering parameters) and/or the reflection coefficients, or an intermediate value used to obtain the reflection coefficients.

In examples, the adjustment (and/or the reduction of the impulse response energy) may be (or be in terms of):

$$25 \quad \gamma = \begin{cases} 1 - (1 - \gamma_{min}) \frac{thresh2 - frameMetrics}{thresh2 - thresh} & , if \text{ frameMetrics} < thresh2 \\ 1 & , otherwise \end{cases}$$

where *thresh2* is the filtering type determination threshold (and may be, for example, 2), *thresh* is the TNS filtering determination threshold (and may be 1.5), γ_{min} is a constant (e.g., a value between 0.7 and 0.95, such as between 0.8 and

0.9, such as 0.85). γ values may be used to scale the LPC coefficients (or other filtering parameters) and/or the reflection coefficients. *frameMetrics* is the frame metrics.

In one example, the formula may be

$$5 \quad \gamma = \begin{cases} 1 - (1 - \gamma_{min}) \frac{thresh2 - predGain}{thresh2 - thresh} & , if \text{ predGain} < thresh2 \\ 1 & , otherwise \end{cases}$$

where *thresh2* is the filtering type determination threshold (and may be, for example, 2), *thresh* is the TNS filtering determination threshold (and may be 1.5), γ_{min} is a constant (e.g., a value between 0.7 and 0.95, such as between 0.8 and 0.9, such as 0.85). γ values may be used to scale the LPC coefficients (or other
10 filtering parameters) and/or the reflection coefficients. *predGain* may be the prediction gain, for example.

From the formula it may be seen that a *frameMetrics* (or *predGain*) lower than *thresh2* but close to it (e.g., 1.999) will cause the reduction of impulse response energy to be weak (e.g. $\gamma \cong 1$). Therefore, the lower impulse response energy LP
15 filtering may be one of a plurality of different lower impulse response energy LP filterings, each being characterized by a different adjustment parameter γ , e.g., in accordance to the value of the frame metrics.

In examples of lower impulse response energy LP filtering, different values of the metrics may cause different adjustments. For example, a higher prediction gain
20 may be associated to a higher a higher value of γ , and a lower reduction of the impulse response energy with respect to the first filter. γ may be seen as a linear function dependent from *predGain*. An increment of *predGain* will cause an increment of γ , which in turn will diminish the reduction of the impulse response energy. If *predGain* is reduced, γ is also reduced, and the impulse response
25 energy will be accordingly also reduced.

Therefore, subsequent frames of the same signal may be differently filtered:

- some frames may be filtered using the first filter (higher impulse response energy filtering), in which the filter parameters (14) are maintained;
- some other frames may be filtered using the second filter (lower impulse response energy filtering), in which the first filter is modified to obtain a second filter with lower impulse response energy (the filter parameters 14 being modified, for example) to reduce the impulse response energy 'with respect to the first filter;
- some other frames may also be filtered using the second filter (lower impulse response energy filtering), but with different adjustment (as a consequence of a different values of the frame metrics).

Accordingly, for each frame, a particular first filter may be defined (e.g., on the basis of the filter parameters), while a second filter may be developed by modifying the filter parameters of the first filter.

Fig. 3A shows an example of the controller 12 and the TNS block 11 cooperating to perform TNS filtering operations.

A frame metrics (e.g., prediction gain) 17 may be obtained and compared to a TNS filtering determination threshold 18a (e.g., at a comparer 10a). If the frame metrics 17 is greater than the TNS filtering determination threshold 18a (thresh), it is permitted (e.g., by the selector 11a) to compare the frame metrics 17 with a filtering type determination threshold 18b (e.g., at a comparer 12a). If the frame metrics 17 is greater than the filtering type determination threshold 18b, then a first filter 14a whose impulse response has higher energy (e.g. $\gamma = 1$) is activated. If the frame metrics 17 is lower than the filtering type determination threshold 18b, then a second filter 15a whose impulse response has lower energy (e.g., $\gamma < 1$) is activated (element 12b indicates a negation of the binary value output by the comparer 12a). The first filter 14a whose impulse response has higher energy may perform filtering S36 with higher impulse response energy, and the second filter 15a whose impulse response has lower energy may perform filtering S35 with lower impulse response energy.

Figs. 3B and 3C shows methods 36 and 35 for using the first and the second filters 14a and 15a, respectively (e.g., for steps S36 and S35, respectively).

The method 36 may comprise a step S36a of obtaining the filter parameters 14. The method 36 may comprise a step S36b performing filtering (e.g., S36) using
 5 the parameters of the first filter 14a. Step S35b may be performed only at the determination (e.g., at step S34) that the frame metrics is over the filtering type determination threshold (e.g., at step S35).

The method 35 may comprise a step S35a of obtaining the filter parameters 14 of the first filter 14a. The method 35 may comprise a step S35b of defining the
 10 adjustment factor γ (e.g., by using at least one of the thresholds thresh and thresh2 and the frame metrics). The method 35 may comprise a step 35c for modifying the first filter 14a to obtain a second filter 15a having lower impulse response energy with respect to the first filter 14a. In particular, the first filter 14a may be modified by applying the adjustment factor γ (e.g., as obtained at S35b) to
 15 the parameters 14 of the first filter 14a, to obtain the parameters of the second filter. The method 35 may comprise a step S35d in which the filtering with the second filter (e.g., at S35 of the method 30) is performed. Steps S35a, S35b, and S35c may be performed at the determination (e.g., at step S34) that the frame metrics is less than the filtering type determination threshold (e.g., at step S35).

20 Fig. 4 shows a method 40' (encoder side) and a method 40'' (decoder side) which may form a single method 40. The methods 40' and 40'' may have some contact in that a decoder operating according to the method 40' may transmit a bitstream (e.g., wirelessly, e.g., using Bluetooth) to a decoder operating according to the method 40''.

25 The steps of method 40 (indicated as a sequence a)-b)-c)-d)-1)-2)-3)-e)-f) and by the sequence S41'-S49') is discussed here below.

a) **Step S41'**: The autocorrelation of the MDCT (or MDST) spectrum (FD value) may be processed, for example,

$$r(k) = \sum_{n=n_{start}}^{n_{stop}-k} c(n)c(n+k), \quad k = 0, \dots, K$$

where K is the LP filter order (e.g. $K = 8$). Here, $c(n)$ may be the FD value input to the TNS tool 11. For example, $c(n)$ may refer to a bin associated to a frequency with index n .

- 5 **b) Step S42':** The autocorrelation may be lag windowed:

$$r(k) = r(k)w(k), \quad k = 0, \dots, K$$

An example of lag windowing function may be, for example:

$$w(k) = \exp\left[-\frac{1}{2}(2\pi\alpha k)^2\right], \quad k = 0, \dots, K$$

where α is a window parameter (e.g. $\alpha = 0.011$).

- 10 **c) Step S43':** LP filter coefficients may be estimated, using e.g. a Levinson-Durbin recursion procedure, such as:

$$e(0) = r(0)$$

$$a^0(0) = 1$$

- 20 for $k = 1$ to K do

$$rc(k) = \frac{-\sum_{n=0}^{k-1} a^{k-1}(n)r(k-n)}{e(k-1)}$$

$$a^k(k) = rc(k)$$

$$a^k(0) = 1$$

- 15 for $n = 1$ to $k - 1$ do

$$a^k(n) = a^{k-1}(n) + rc(k)a^{k-1}(k-n)$$

$$e(k) = (1 - rc(k)^2)e(k-1)$$

where $a(k) = a^k(k)$, $k = 0, \dots, K$ are the estimated LPC coefficients (or other filtering parameters), $rc(k)$, $k = 1, \dots, K$ are the corresponding reflection coefficients and $e = e(K)$ is the prediction error.

- d) **Step S44'**: The decision (step S44' or S32) to turn on/off TNS filtering in the current frame may be based on e.g. a frame metrics, such as the prediction gain:

If $predGain > thresh$, then turn on TNS filtering

5 where the prediction gain is computed by

$$predGain = \frac{r(0)}{e}$$

and $thresh$ is a threshold (e.g. $thresh = 1.5$).

- 1) **Step S45'**: The weighting factor γ may be obtained (e.g., at step S45') by

$$\gamma = \begin{cases} 1 - (1 - \gamma_{min}) \frac{thresh2 - predGain}{thresh2 - thresh} & , \text{ if } predGain < thresh2 \\ 1 & , \text{ otherwise} \end{cases}$$

10 where $thresh2$ is a second threshold (e.g. $thresh2 = 2$) and γ_{min} is the minimum weighting factor (e.g. $\gamma_{min} = 0.85$). The $thresh2$ may be, for example, the filtering type determination threshold.

When $\gamma = 1$, the first filter 14a is used. When $0 < \gamma < 1$, the second filter 15a is used (e.g., at step S35b).

- 15 2) **Step S46'**: The LPC coefficients (or other filtering parameters) may be weighted (e.g., at step S46') using the factor γ :

$$a_w(k) = \gamma^k a(k), k = 0, \dots, K$$

γ^k is an exponentiation (e.g., $\gamma^2 = \gamma * \gamma$).

- 20 3) **Step S47'**: The weighted LPC coefficients (or other filtering parameters) may be converted to reflection coefficients using, e.g., the following procedure (step S47'):

$$a^K(k) = a_w(k), k = 0, \dots, K$$

for $k = K$ to 1 do

$$rc(k) = a^k(k)$$

$$e = (1 - rc(k)^2)$$

for $n = 1$ to $k - 1$ do

$$a^{k-1}(n) = \frac{a^k(n) - rc(k)a^k(k-n)}{e}$$

- 5 **e) Step S48'** :If TNS is on (as a result of the determination of at S32, for example), the reflection coefficients may be quantized (step S48') using, e.g., scalar uniform quantization in the arcsine domain:

$$rc_i(k) = \text{round} \left[\frac{\arcsin(rc(k))}{\Delta} \right]$$

$$rc_q(k) = \sin(\Delta rc_i(k))$$

- 10 where Δ is the cell width (e.g. $\Delta = \frac{\pi}{17}$) and $\text{round}(\cdot)$ is the rounding-to-nearest-integer function.
 $rc_i(k)$ are the quantizer output indices which are then encoded using e.g. arithmetic encoding.
 $rc_q(k)$ are the quantized reflection coefficients.

- 15 **f) Step S49'**: If TNS is on, the MDCT (or MDST) spectrum is filtered (step S49') using the quantized reflection coefficients and a lattice filter structure

$$s^0(n_{start} - 1) = s^1(n_{start} - 1) = \dots = s^{K-1}(n_{start} - 1) = 0$$

for $n = n_{start}$ to n_{stop} do

$$t^0(n) = s^0(n) = c(n)$$

for $k = 1$ to K do

$$t^k(n) = t^{k-1}(n) + rc_q(k)s^{k-1}(n-1)$$

20 $s^k(n) = rc_q(k)t^{k-1}(n) + s^{k-1}(n-1)$

$$c_f(n) = t^K(n)$$

- A bitstream may be transmitted to the decoder. The bitstream may comprise,
 25 together with an FD representation of the information signal (e.g., an audio signal), also control data, such as the reflection coefficients obtained by performing TNS operations described above (TNS analysis).

The method 40" (decoder side) may comprise steps g) (S41") and h) (S42") in which, if TNS is on, the quantized reflection coefficients are decoded and the quantized MDCT (or MDST) spectrum is filtered back. The following procedure may be used:

```

10   $s^0(n_{start} - 1) = s^1(n_{start} - 1) = \dots = s^{K-1}(n_{start} - 1) = 0$ 
    for  $n = n_{start}$  to  $n_{stop}$  do
5       $t^K(n) = c(n)$ 
      for  $k = K$  to 1 do
         $t^{k-1}(n) = t^k(n) - rc_q(k)s^{k-1}(n-1)$ 
         $s^k(n) = rc_q(k)t^{k-1}(n) + s^{k-1}(n-1)$ 
       $c_f(n) = s^0(n) = t^0(n)$ 

```

An example of encoder apparatus 50 (which may embody the encoder apparatus 10 and/or perform at least some of the operation of the methods 30 and 40') is shown in Fig. 5.

```

15  The encoder apparatus 50 may comprise a plurality of tools for encoding an input
    signal (which may be, for example, an audio signal). For example, a MDCT tool 51
    may transform a TD representation of an information signal to an FD
    representation. A spectral noise shaper, SNS, tool 52 may perform noise shaping
    analysis (e.g., a spectral noise shaping, SNS, analysis), for example, and retrieve
20  LPC coefficients or other filtering parameters (e.g.,  $a(k)$ , 14). The TNS tool 11 may
    be as above and may be controlled by the controller 12. The TNS tool 11 may
    perform a filtering operation (e.g. according to method 30 or 40') and output both a
    filtered version of the information signal and a version of the reflection coefficients.
    A quantizer tool 53 may perform a quantization of data output by the TNS tool 11.
25  An arithmetic coder 54 may provide, for example, entropy coding. A noise level
    tool 55' may also be used for estimating a noise level of the signal. A bitstream
    writer 55 may generate a bitstream associated to the input signal that may be
    transmitted (e.g., wireless, e.g., using Bluetooth) and/or stored.

```

A bandwidth detector 58' (which may detect the bandwidth of the input signal) may also be used. It may provide the information on active spectrum of the signal. This information may also be used, in some examples, to control the coding tools.

5 The encoder apparatus 50 may also comprise a long term post filtering tool 57 which may be input with a TD representation of the input signal, e.g., after that the TD representation has been downsampled by a downsampler tool 56.

An example of decoder apparatus 60 (which may embody the decoder apparatus 20 and/or perform at least some of the operation of the method 40") is shown in Fig. 6.

10 The decoder apparatus 60 may comprise a reader 61 which may read a bitstream (e.g., as prepared by the apparatus 50). The decoder apparatus 60 may comprise an arithmetic residual decoder 61a which may perform, for example, entropy decoding, residual decoding, and/or arithmetic decoding with a digital representation in the FD (restored spectrum), e.g., as provided by the decoder.

15 The decoder apparatus 60 may comprise a noise filing tool 62 and a global gain tool 63, for example. The decoder apparatus 60 may comprise a TNS decoder 21 and a TNS decoder controller 22. The apparatus 60 may comprise an SNS decoder tool 65, for example. The decoder apparatus 60 may comprise an inverse MDCT (or MDST) tool 65' to transform a digital representation of the information

20 signal from the FD to the TD. A long term post filtering may be performed by the LTPF tool 66 in the TD. Bandwidth information 68 may be obtained from the bandwidth detector 58', for example, and applied to some of the tools (e.g., 62 and 21).

Examples of the operations of the apparatus above are here provided.

25 Temporal Noise Shaping (TNS) may be used by tool 11 to control the temporal shape of the quantization noise within each window of the transform.

In examples, if TNS is active in the current frame, up to two filters per MDCT-spectrum (or MDST spectrum or other spectrum or other FD representation) may

be applied. It is possible to apply a plurality of filters and/or to perform TNS filtering on a particular frequency range. In some examples, this is only optional.

The number of filters for each configuration and the start and the stop frequency of each filter are given in the following table:

Band-width	num_tns_filters	start_freq(f)	stop_freq(f)	sub_start(f,s)	sub_stop(f,s)
NB	1	{12}	{80}	{{12,34,57}}	{{34,57,80}}
WB	1	{12}	{160}	{{12,61,110}}	{{61,110,160}}
SSWB	1	{12}	{240}	{{12,88,164}}	{{88,164,240}}
SWB	2	{12,160}	{160,320}	{{12,61,110}, {160,213,266}}	{{61,110,160}, {213,266,320}}
FB	2	{12,200}	{200,400}	{{12,74,137}, {200,266,333}}	{{74,137,200}, {266,333,400}}

Information such as the start and stop frequencies may be signalled, for example, from the bandwidth detector 58'.

Where NB is narrowband, WB is wideband, SSWB is semi-super wideband, SWB
5 is super wideband, and FB is full wideband.

The TNS encoding steps are described in the below. First, an analysis may estimate a set of reflection coefficients for each TNS filter. Then, these reflection coefficients may be quantized. And finally, the MDCT-spectrum (or MDST spectrum or other spectrum or other FD representation) may be filtered using the
10 quantized reflection coefficients.

The complete TNS analysis described below is repeated for every TNS filter f , with $f = 0 \dots \text{num_tns_filters}-1$ (num_tns_filters being provided by the table above).

A normalized autocorrelation function may be calculated (e.g., at step S41') as follows, for each $k = 0 \dots 8$

$$r(k) = \begin{cases} r_0(k) & , \text{ if } \prod_{s=0}^2 e(s) = 0 \\ \sum_{s=0}^2 \frac{\sum_{n=\text{sub_start}(f,s)}^{\text{sub_stop}(f,s)-1-k} X_s(n)X_s(n+k)}{e(s)} & , \text{ otherwise} \end{cases}$$

with

$$r_0(k) = \begin{cases} 1 & , \text{ if } k = 0 \\ 0 & , \text{ otherwise} \end{cases}$$

and

$$5 \quad e(s) = \sum_{n=\text{sub_start}(f,s)}^{\text{sub_stop}(f,s)-1} X_s(n)^2 \quad \text{for } s = 0..2$$

with $\text{sub_start}(f, s)$ and $\text{sub_stop}(f, s)$ are given in the table above.

The normalized autocorrelation function may be lag-windowed (e.g., at S42') using, for example:

$$r(k) = r(k) \exp \left[-\frac{1}{2} (0.02\pi k)^2 \right] \quad \text{for } k = 0..8$$

- 10 The Levinson-Durbin recursion described above may be used (e.g., at step S43') to obtain LPC coefficients or other filtering parameters $a(k)$, $k = 0..8$ and/or a prediction error e .

The decision to turn on/off the TNS filter f in the current frame is based on the prediction gain:

- 15 If $\text{predGain} > \text{thresh}$, then turn on the TNS filter f

With, for example, $\text{thresh} = 1.5$ and the prediction gain being obtained, for example, as:

$$\text{predGain} = \frac{r(0)}{e}$$

The additional steps described below are performed only if the TNS filter f is turned on (e.g., if the step S32 has result "YES").

A weighting factor γ is computed by

$$\gamma = \begin{cases} 1 - (1 - \gamma_{min}) \frac{thresh2 - predGain}{thresh2 - thresh} & , \text{if } predGain < thresh2 \\ 1 & , \text{otherwise} \end{cases}$$

5 with $thresh2 = 2$, $\gamma_{min} = 0.85$ and

$$tns_lpc_weighting = \begin{cases} 1 & , \text{if } nbits < 480 \\ 0 & , \text{otherwise} \end{cases}$$

The LPC coefficients or other filtering parameters may be weighted (e.g., at step S46') using the factor γ

$$a_w(k) = \gamma^k a(k) \quad \text{for } k = 0..8$$

10 The weighted LPC coefficients or other filtering parameters may be converted (e.g., at step S47') to reflection coefficients using, for example, the following algorithm:

$$a^K(k) = a_w(k), k = 0, \dots, K$$

for $k = K$ to 1 do

$$rc(k) = a^k(k)$$

$$e = (1 - rc(k)^2)$$

15 for $n = 1$ to $k - 1$ do

$$a^{k-1}(n) = \frac{a^k(n) - rc(k)a^k(k-n)}{e}$$

wherein $rc(k, f) = rc(k)$ are the final estimated reflection coefficients for the TNS
20 filter f .

If the TNS filter f is turned off (e.g., outcome "NO" at the check of step S32), then the reflection coefficients may be simply set to 0: $rc(k, f) = 0$, $k = 0..8$.

The quantization process, e.g., as performed at step S48', is now discussed.

For each TNS filter f , the reflection coefficients obtained may be quantized, e.g., using scalar uniform quantization in the arcsine domain

$$rc_i(k, f) = \text{nint} \left[\frac{\arcsin(rc(k, f))}{\Delta} \right] + 8 \quad \text{for } k = 0..8$$

and

$$5 \quad rc_q(k, f) = \sin[\Delta(rc_i(k, f) - 8)] \quad \text{for } k = 0..8$$

wherein $\Delta = \frac{\pi}{17}$ and $\text{nint}(\cdot)$ is the rounding-to-nearest-integer function, for example.

$rc_i(k, f)$ may be the quantizer output indices and $rc_q(k, f)$ may be the quantized reflection coefficients.

The order of the quantized reflection coefficients may be calculated using

$$10 \quad k = 7$$

while $k \geq 0$ and $rc_q(k, f) = 0$ do

$$k = k - 1$$

$$rc_{order}(f) = k + 1$$

The total number of bits consumed by TNS in the current frame can then be

15 computed as follows

$$nbits_{TNS} = \left\lceil \sum_{f=0}^{\text{num_tns_filters}-1} \frac{2048 + nbits_{TNS_{order}}(f) + nbits_{TNS_{rc}}(f)}{2048} \right\rceil$$

with

$$nbits_{TNS_{order}}(f) = \begin{cases} \text{ac_tns_order_bits}[\text{tns_lpc_weighting}][rc_{order}(f) - 1] & , \text{if } rc_{order}(f) > 0 \\ 0 & , \text{otherwise} \end{cases}$$

and

$$20 \quad nbits_{TNS_{coef}}(f) = \begin{cases} \sum_{k=0}^{rc_{order}(f)-1} \text{ac_tns_coef_bits}[k][rc_i(k, f)] & , \text{if } rc_{order}(f) > 0 \\ 0 & , \text{otherwise} \end{cases}$$

The values of `tab_nbits_TNS_order` and `tab_nbits_TNS_coef` may be provided in tables.

The MDCT (or MDST) spectrum $X_s(n)$ (input 15 in Fig. 1) may be filtered using the following procedure:

$$5 \quad s^0(\text{start_freq}(0) - 1) = s^1(\text{start_freq}(0) - 1) = \dots = s^7(\text{start_freq}(0) - 1) = 0$$

for $f = 0$ to `num_tns_filters`-1 do

for $n = \text{start_freq}(f)$ to `stop_freq(f) - 1` do

$$t^0(n) = s^0(n) = X_s(n)$$

for $k = 0$ to 7 do

$$t^{k+1}(n) = t^k(n) + rc_q(k)s^k(n-1)$$

$$10 \quad s^{k+1}(n) = rc_q(k)t^k(n) + s^k(n-1)$$

$$X_f(n) = t^8(n)$$

wherein $X_f(n)$ is the TNS filtered MDCT (or MDST) spectrum (output 15 in Fig. 1).

With reference to operations performed at the decoder (e.g., 20, 60), quantized

15 reflection coefficients may be obtained for each TNS filter f using

$$rc_q(k, f) = \sin[\Delta(rc_i(k, f) - 8)] \quad k = 0..8$$

wherein $rc_q(k, f)$ are the quantizer output indices.

The MDCT (or MDST) spectrum $\widehat{X}_f(n)$ as provided to the TNS decoder 21 (e.g., as obtained from the global gain tool 63) may then be filtered using the following

20 algorithm

$$s^0(\text{start_freq}(0) - 1) = s^1(\text{start_freq}(0) - 1) = \dots = s^7(\text{start_freq}(0) - 1) = 0$$

for $f = 0$ to `num_tns_filters`-1 do

for $n = \text{start_freq}(f)$ to `stop_freq(f) - 1` do

$$t^K(n) = \widehat{X}_f(n)$$

for $k = 7$ to 0 do

$$25 \quad t^k(n) = t^{k+1}(n) - rc_q(k)s^k(n-1)$$

$$s^{k+1}(n) = rc_q(k)t^k(n) + s^k(n-1)$$

$$\widehat{X}_s(n) = s^0(n) = t^0(n)$$

wherein $\widehat{X}_s(n)$ is the output of the TNS decoder.

6. Discussion on the invention

- 5 As explained above, TNS can sometimes introduce artefacts, degrading the quality of the audio coder. These artefacts are click-like or noise-like and appear in most of the cases with speech signals or tonal music signals.

It was observed that artefacts generated by TNS only occur in frames where the prediction gain predGain is low and close to a threshold thresh.

- 10 One could think that increasing the threshold would easily solve the problem. But for most of the frames, it is actually beneficial to turn on TNS even when the prediction gain is low.

Our proposed solution is to keep the same threshold but to adjust the TNS filter when the prediction gain is low, so as to reduce the impulse response energy.

- 15 There are many ways to implement this adjustment (which in some cases may be referred to as "attenuation", e.g., when the reduction of impulse response energy is obtained by reducing the LP filter parameters, for example). We may choose to use weighting, which may be, for example, a weighting

$$a_w(k) = \gamma^k a(k), k = 0, \dots, K$$

- 20 with $a(k)$ are the LP filter parameters (e.g., LPC coefficients) computed in Encoder Step c) and $a_w(k)$ are the weighted LP filter parameters. The adjustment (weighting) factor γ is made dependent on the prediction gain such that higher reduction of impulse response energy ($\gamma < 1$) is applied for lower prediction gains and such that there is, for example, no reduction of impulse response energy ($\gamma =$
25 1) for higher prediction gains.

The proposed solution was proven to be very effective at removing all artefacts on problematic frames while minimally affecting the other frames.

Reference can now be made to Figs. 8(1)-8(3). The figures show a frame of audio signal (continuous line) and the frequency response (dashed line) of the
5 corresponding TNS prediction filter.

Fig. 8(1): castanets signal

Fig. 8(2): pitch pipe signal

Fig. 8(3): speech signal

The prediction gain is related to the flatness of the signal's temporal envelope
10 (see, for example, Section 3 of ref [2] or Section 1.2 of ref [3]).

A low prediction gain implies a tendentially flat temporal envelope, while a high prediction gain implies an extremely un-flat temporal envelope.

Figure 8(1) shows the case of a very low prediction gain ($\text{predGain}=1.0$). It corresponds to the case of a very stationary audio signal, with a flat temporal
15 envelope. In this case $\text{predGain} = 1 < \text{thresh}$ (e.g., $\text{thresh}=1.5$): no filtering is performed (S33).

Figure 8(2) shows the case of a very high prediction gain (12.3). It corresponds to the case of a strong and sharp attack, with a highly un-flat temporal envelope. In this case $\text{predGain} = 12.3 > \text{thresh2}$ ($\text{thresh2}=2$): higher impulse response energy
20 filtering is performed at S36.

Figure 8(3) shows the case of a prediction gain between thresh and thresh2 , e.g., in a 1.5-2.0 range (higher than the first threshold, lower than the second threshold). It corresponds to the case of a slightly un-flat temporal envelope. In this case $\text{thresh} < \text{predGain} < \text{thresh2}$: lower impulse response energy filtering is
25 performed at S35, using the second filter 15a with lower impulse response energy.

7. Other examples

Fig. 7 shows an apparatus 110 which may implement the encoding apparatus 10 or 50 and/or perform at least some steps of the method 30 and/or 40'. The apparatus 110 may comprise a processor 111 and a non-transitory memory unit 112 storing instructions which, when executed by the processor 111, may cause the processor 111 to perform a TNS filtering and/or analysis. The apparatus 110 may comprise an input unit 116, which may obtain an input information signal (e.g., an audio signal). The processor 111 may therefore perform TNS processes.

Fig. 8 shows an apparatus 120 which may implement the decoder apparatus 20 or 60 and/or perform the method 40'. The apparatus 120 may comprise a processor 121 and a non-transitory memory unit 122 storing instructions which, when executed by the processor 121, may cause the processor 121 to perform, inter alia, a TNS synthesis operation. The apparatus 120 may comprise an input unit 126, which may obtain a decoded representation of an information signal (e.g., an audio signal) in the FD. The processor 121 may therefore perform processes to obtain a decoded representation of the information signal, e.g., in the TD. This decoded representation may be provided to external units using an output unit 127. The output unit 127 may comprise, for example, a communication unit to communicate to external devices (e.g., using wireless communication, such as Bluetooth) and/or external storage spaces. The processor 121 may save the decoded representation of the audio signal in a local storage space 128.

In examples, the systems 110 and 120 may be the same device.

Depending on certain implementation requirements, examples may be implemented in hardware. The implementation may be performed using a digital storage medium, for example a floppy disk, a Digital Versatile Disc (DVD), a Blu-Ray Disc, a Compact Disc (CD), a Read-only Memory (ROM), a Programmable Read-only Memory (PROM), an Erasable and Programmable Read-only Memory (EPROM), an Electrically Erasable Programmable Read-Only Memory (EEPROM) or a flash memory, having electronically readable control signals stored thereon, which cooperate (or are capable of cooperating) with a programmable computer system such that the respective method is performed.

Therefore, the digital storage medium may be computer readable.

Generally, examples may be implemented as a computer program product with program instructions, the program instructions being operative for performing one of the methods

when the computer program product runs on a computer. The program instructions may for example be stored on a machine readable medium.

Other examples comprise the computer program for performing one of the methods described herein, stored on a machine readable carrier. In other words, an example of
5 method is, therefore, a computer program having a program instructions for performing one of the methods described herein, when the computer program runs on a computer.

A further example of the methods is, therefore, a data carrier medium (or a digital storage medium, or a computer-readable medium) comprising, recorded thereon, the computer
10 program for performing one of the methods described herein. The data carrier medium, the digital storage medium or the recorded medium are tangible and/or non-transitionary, rather than signals which are intangible and transitory.

A further example comprises a processing unit, for example a computer, or a programmable logic device performing one of the methods described herein.

A further example comprises a computer having installed thereon the computer program
15 for performing one of the methods described herein.

A further example comprises an apparatus or a system transferring (for example, electronically or optically) a computer program for performing one of the methods described herein to a receiver. The receiver may, for example, be a computer, a mobile device, a memory device or the like. The apparatus or system may, for example, comprise
20 a file server for transferring the computer program to the receiver.

In some examples, a programmable logic device (for example, a field programmable gate array) may be used to perform some or all of the functionalities of the methods described herein. In some examples, a field programmable gate array may cooperate with a microprocessor in order to perform one of the methods described herein. Generally, the
25 methods may be performed by any appropriate hardware apparatus.

The above described examples are illustrative for the principles discussed above. It is understood that modifications and variations of the arrangements and the details described herein will be apparent. It is the intent, therefore, to be limited by the scope of the impending patent claims and not by the specific details presented by way of
30 description and explanation of the examples herein.

Claims

1. An encoder apparatus comprising:

a temporal noise shaping, TNS, tool for performing linear prediction, LP, filtering on an information signal including a plurality of frames; and

5 a controller configured to control the TNS tool so that the TNS tool performs LP filtering with:

a first filter whose impulse response has a higher energy; and

a second filter whose impulse response has a lower energy, wherein the second filter is not an identity filter,

10 wherein the controller is configured to choose between filtering with the first filter and filtering with the second filter on the basis of a frame metrics,

wherein the controller is further configured to: modify the first filter so as to obtain the second filter in which the filter's impulse response energy is reduced.

2. The encoder apparatus of claim 1, wherein the controller is further configured to:

15 apply an adjustment factor to the first filter to obtain the second filter.

3. The encoder apparatus of claim 2, configured to modify the first filter to obtain the second filter by modifying the amplitude of the parameters of the first filter using an adjustment factor.

4. The encoder apparatus of claim 2 or 3, wherein the controller is further configured
20 to:

define the adjustment factor on the basis of a filtering type determination threshold used for selecting between filtering with the first filter and filtering with the second filter.

5. The encoder apparatus of claim 2 or 3 or 4, wherein the controller is further configured to:

25 define the adjustment factor on the basis of at least the frame metrics.

6. The encoder apparatus of any of claims 2-5, wherein the controller is further configured to:

define the adjustment factor on the basis of a TNS filtering determination threshold which is used for selecting between performing TNS filtering and non-performing TNS filtering.

7. The encoder apparatus of any of claims 2-6, wherein the controller is further configured to:

define the adjustment factor using a linear function of the frame metrics, the linear function being such that an increase in the frame metrics corresponds to an increase of the adjustment factor and/or of the filter's impulse response energy.

8. The encoder apparatus of any of claims 2-7, configured to define the adjustment factor as

$$\gamma = \begin{cases} 1 - (1 - \gamma_{\min}) \frac{\text{thresh2} - \text{frameMetrics}}{\text{thresh2} - \text{thresh}} & , \text{ if } \text{frameMetrics} < \text{thresh2} \\ 1 & , \text{ otherwise} \end{cases}$$

wherein thresh is the TNS filtering determination threshold, thresh2 is the filtering type determination threshold, frameMetrics is a frame metrics, and $\gamma_{\min}$ is a fixed value.

9. The encoder apparatus of any of claims 2-8, configured to modify the parameters of the first filter to obtain the parameters of the second filter by applying:

$$\mathbf{a}_w(\mathbf{k}) = \gamma^k \mathbf{a}(\mathbf{k}), \mathbf{k} = 0, \dots, K$$

where $\mathbf{a}(\mathbf{k})$ are parameters of the first filter, γ is the adjustment factor such that $0 < \gamma < 1$, $\mathbf{a}_w(\mathbf{k})$ are the parameters of the second filter and K is the order of the first filter.

10. The encoder apparatus of any of the preceding claims, wherein the controller is further configured to:

obtain the frame metrics from at least one of a prediction gain, an energy of the information signal and/or a prediction error.

11. The encoder apparatus of any of the preceding claims, wherein the frame metrics comprises a prediction gain calculated as

$$predGain = \frac{energy}{predError}$$

where *energy* is a term associated to an energy of the information signal, and *predError* is a term associated to a prediction error.

12. The encoder apparatus of any of the preceding claims, wherein the controller is configured so that:

5 at least for a reduction of a prediction gain and/or a reduction of an energy of the information signal, the second filter's impulse response energy is reduced, and/or at least for an increase of the prediction error, the second filter's impulse response energy is reduced.

10 13. The encoder apparatus of any of the preceding claims, wherein the controller is further configured to:

 compare the frame metrics with a filtering type determination threshold, so as to perform a filtering with the first filter when the frame metrics is lower than the filtering type determination threshold.

15 14. The encoder apparatus of any of the preceding claims, wherein the controller is further configured to:

 choose between performing a filtering and non-performing filtering on the basis of the frame metrics.

15. The encoder apparatus of claim 14, wherein the controller is further configured to:

20 compare the frame metrics with a TNS filtering determination threshold, so as to choose to avoid TNS filtering when the frame metrics is lower than the TNS filtering determination threshold.

16. The encoder apparatus of any of the preceding claims, further comprising:

 a bitstream writer to prepare a bitstream with reflection coefficients, or a quantized version thereof, obtained by the TNS tool.

25 17. The encoder apparatus of any of the preceding claims, the filtering parameters of the first filter being chosen between LP coding, LPC, coefficients and/or any other representation of the filter coefficients.

18. The encoder apparatus of any of the preceding claims, wherein the information signal is an audio signal.

19. The encoder apparatus according to any of the preceding claims, wherein the controller is further configured to modify the first filter so as to obtain the second filter in
5 which the filter's impulse response energy is reduced.

20. The encoder apparatus of any of the preceding claims, wherein the frame metrics is associated to the flatness of the signal's temporal envelope.

21. A method for performing temporal noise shaping, TNS, filtering on an information signal including a plurality of frames, the method comprising:

10 for each frame, choosing between filtering with a first filter and filtering with a second filter, whose impulse response has a lower energy, on the basis of a frame metrics, wherein the second filter is not an identity filter;

filtering the frame using the filtering with the filtering chosen between filtering with the first filter and filtering with the second filter; and

15 modify the first filter so as to obtain the second filter in which the filter's impulse response energy is reduced.

22. A non-transitory storage device storing instructions which, when executed by a processor cause the processor to perform the method of claim 21.