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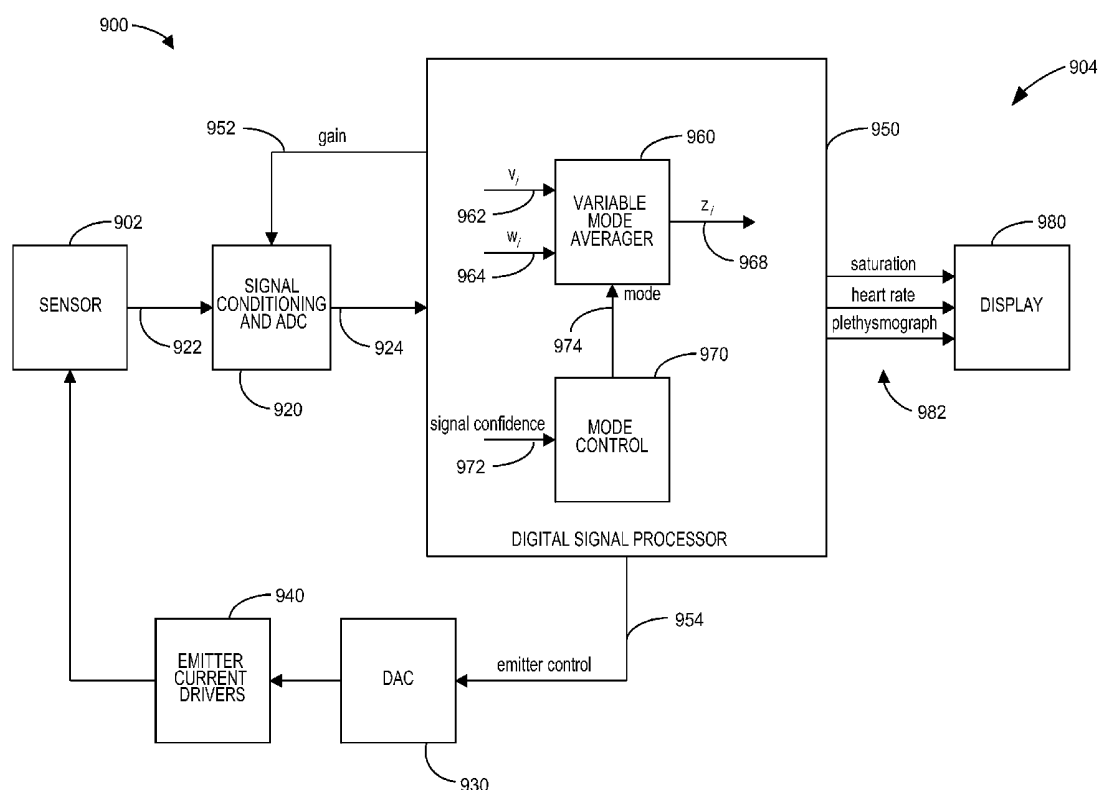
(19) **United States**(12) **Patent Application Publication**  
**Weber et al.**(10) **Pub. No.: US 2018/0256113 A1**(43) **Pub. Date: Sep. 13, 2018**(54) **VARIABLE INDICATION ESTIMATOR****Publication Classification**(71) Applicant: **MASIMO CORPORATION**, Irvine,  
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**Lorenzo Cazzoli**, Barcelona (ES)(21) Appl. No.: **15/913,044**(22) Filed: **Mar. 6, 2018**(51) **Int. Cl.****A61B 5/00** (2006.01)**G06K 9/00** (2006.01)**A61B 5/021** (2006.01)**A61B 5/1455** (2006.01)(52) **U.S. Cl.**CPC ..... **A61B 5/7271** (2013.01); **G06K 9/0051**  
(2013.01); **A61B 5/6829** (2013.01); **A61B**  
**5/021** (2013.01); **A61B 5/14551** (2013.01);  
**A61B 5/7239** (2013.01); **A61B 5/6838**  
(2013.01); **A61B 5/0002** (2013.01); **A61B**  
**5/6826** (2013.01); **A61B 5/1455** (2013.01);  
**A61B 5/7207** (2013.01); **A61B 5/72** (2013.01)**Related U.S. Application Data**

(63) Continuation of application No. 14/830,211, filed on Aug. 19, 2015, now abandoned, which is a continuation of application No. 13/942,562, filed on Jul. 15, 2013, now Pat. No. 9,138,192, which is a continuation of application No. 13/601,930, filed on Aug. 31, 2012, now Pat. No. 8,489,364, which is a continuation of application No. 13/007,109, filed on Jan. 14, 2011, now Pat. No. 8,260,577, which is a continuation of application No. 12/362,463, filed on Jan. 29, 2009, now Pat. No. 7,873,497, which is a continuation of application No. 11/375,662, filed on Mar. 14, 2006, now Pat. No. 7,499,835, which is a continuation of application No. 10/213,270, filed on Aug. 5, 2002, now Pat. No. 6,999,904, which is a continuation-in-part of application No. 09/586,845, filed on Jun. 5, 2000, now Pat. No. 6,430,525.

(57)

**ABSTRACT**

A variable indication estimator which determines an output value representative of a set of input data. For example, the estimator can reduce input data to estimates of a desired signal, select a time, and determine an output value from the estimates and the time. In one embodiment, the time is selected using one or more adjustable signal confidence parameters determine where along the estimates the output value will be computed. By varying the parameters, the characteristics of the output value are variable. For example, when input signal confidence is low, the parameters are adjusted so that the output value is a smoothed representation of the input signal. When input signal confidence is high, the parameters are adjusted so that the output value has a faster and more accurate response to the input signal.



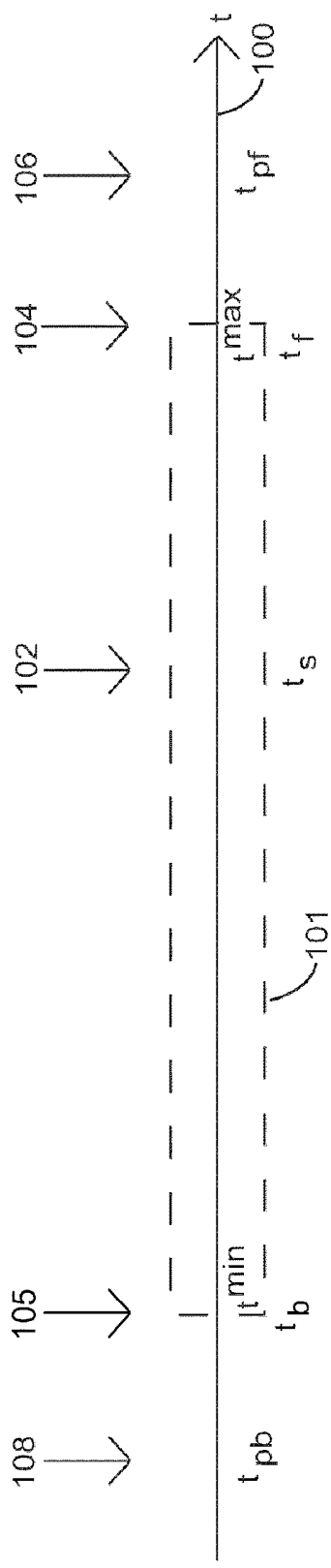


FIG. 1

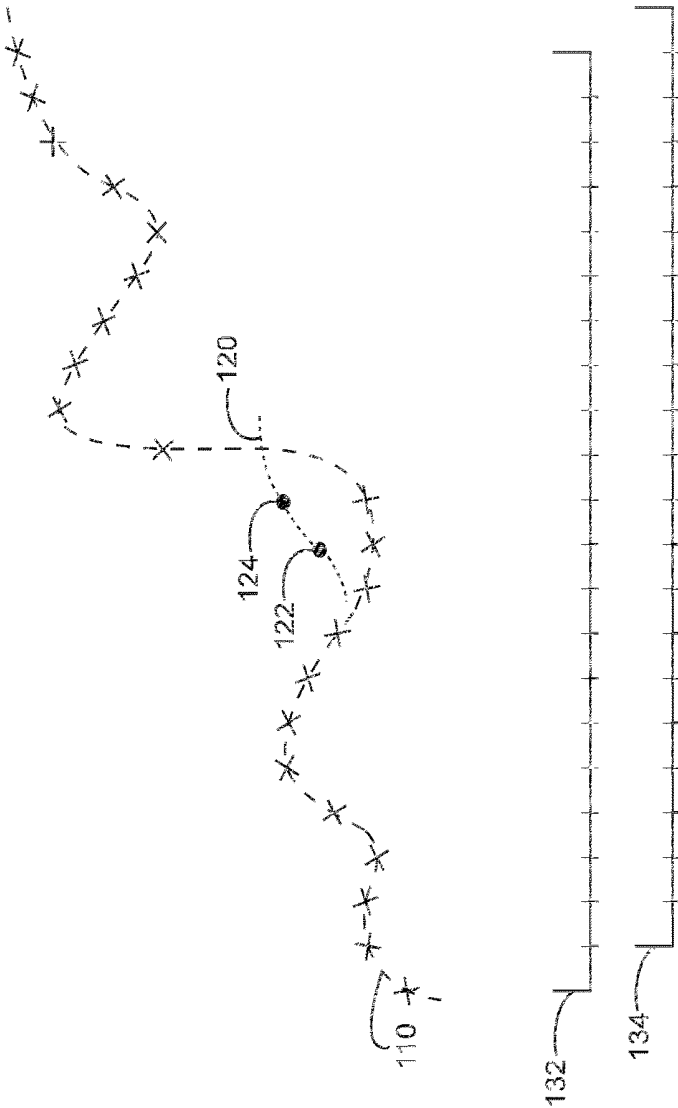


FIG. 2A

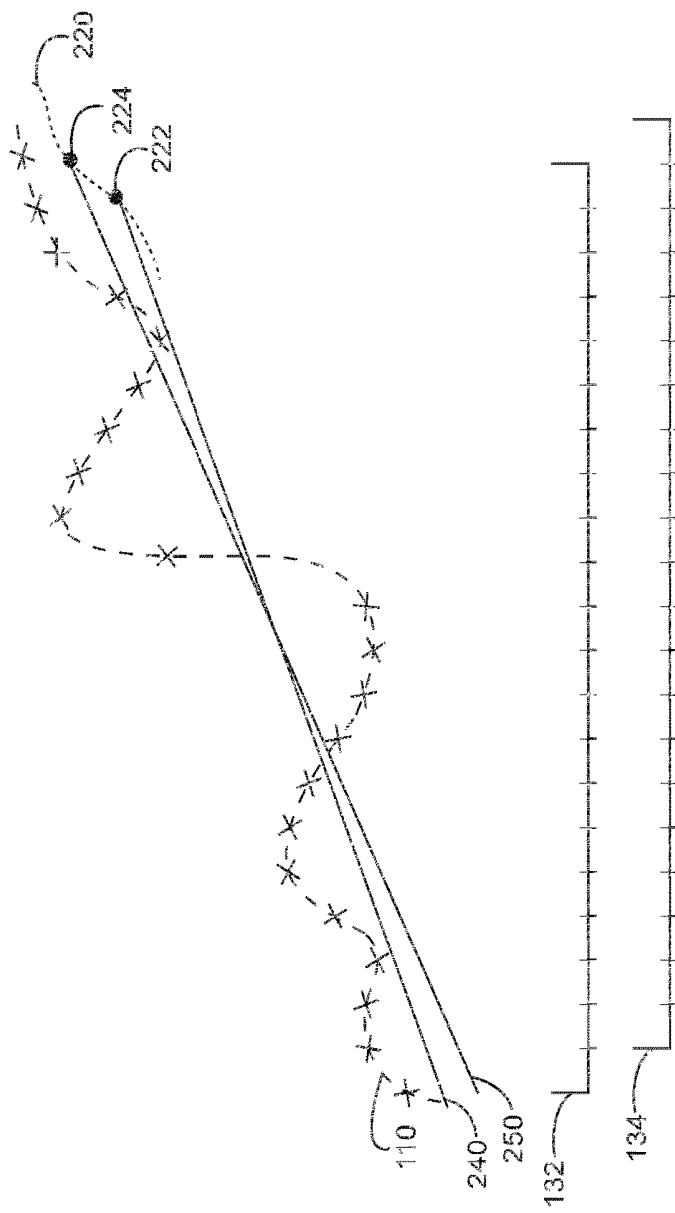


FIG. 2B

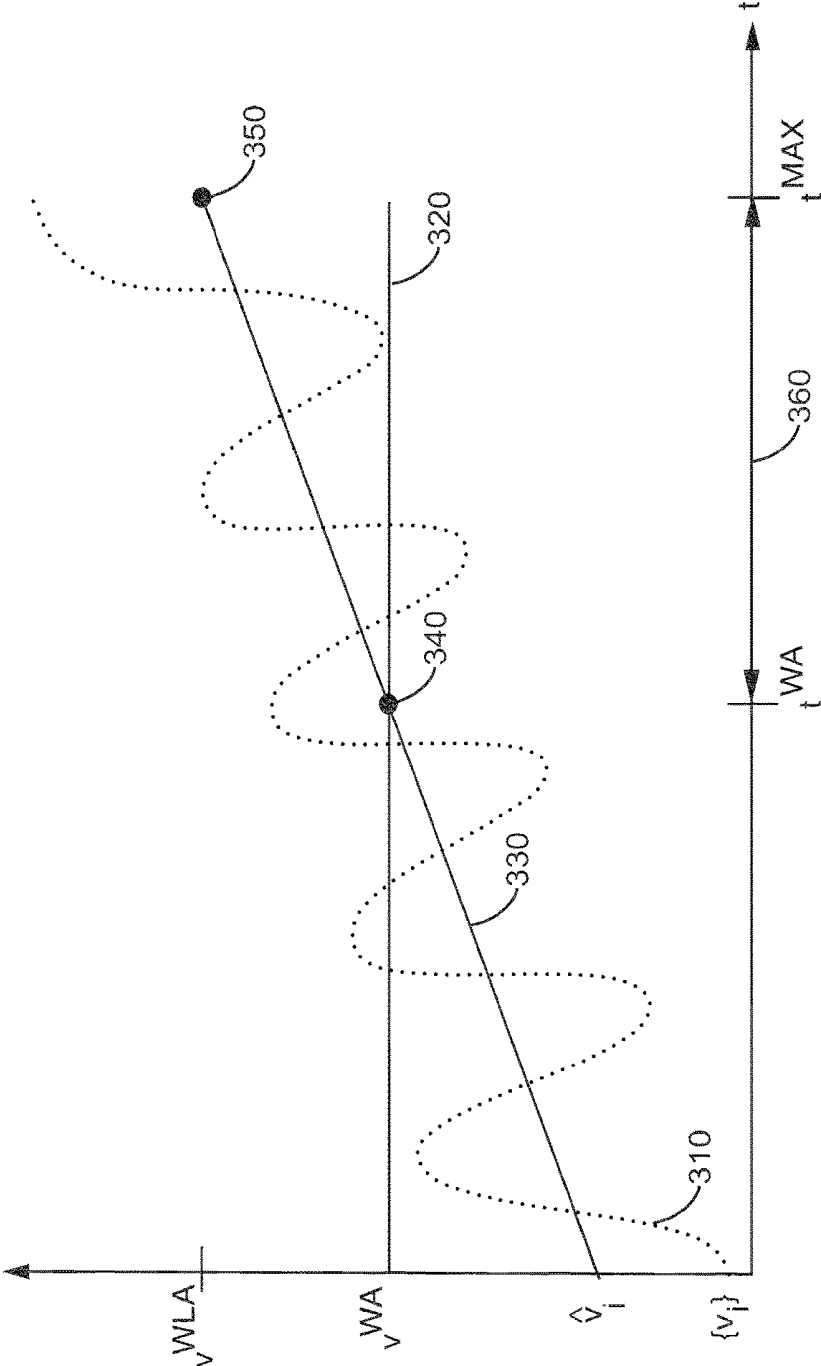


FIG. 3

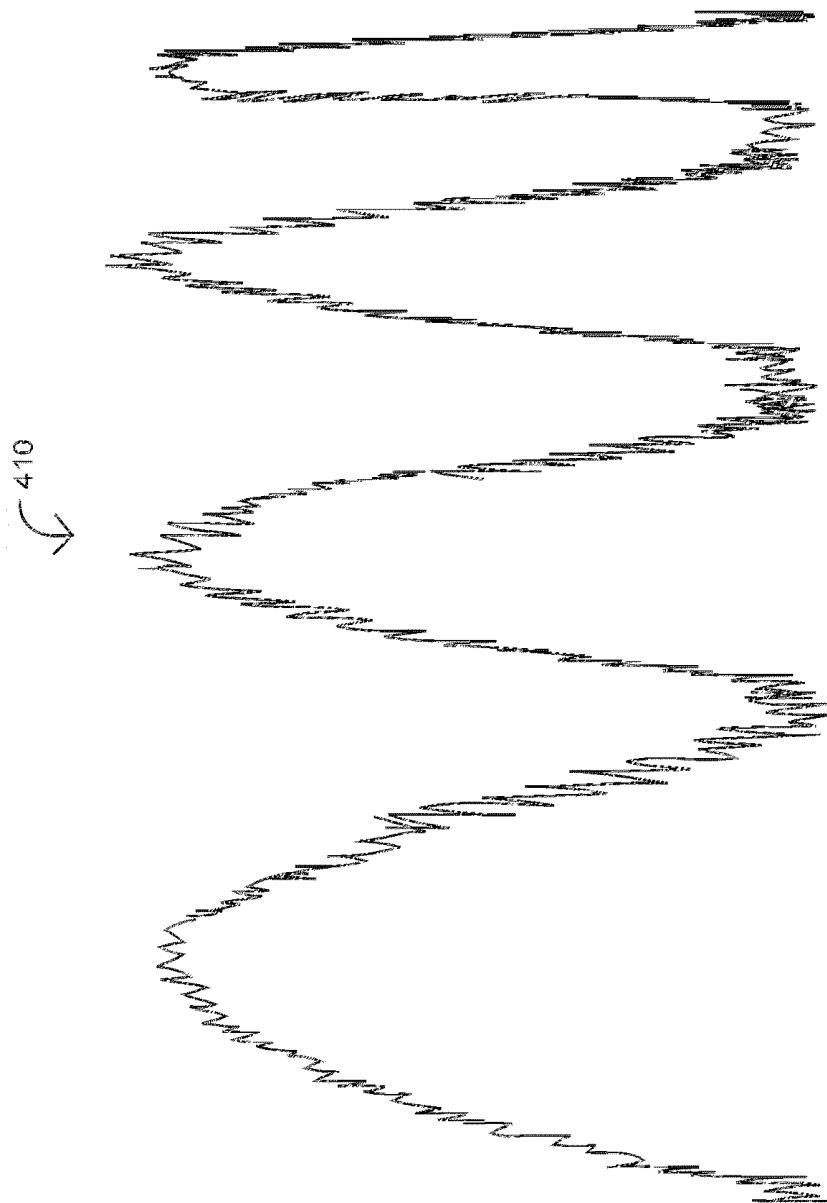


FIG. 4

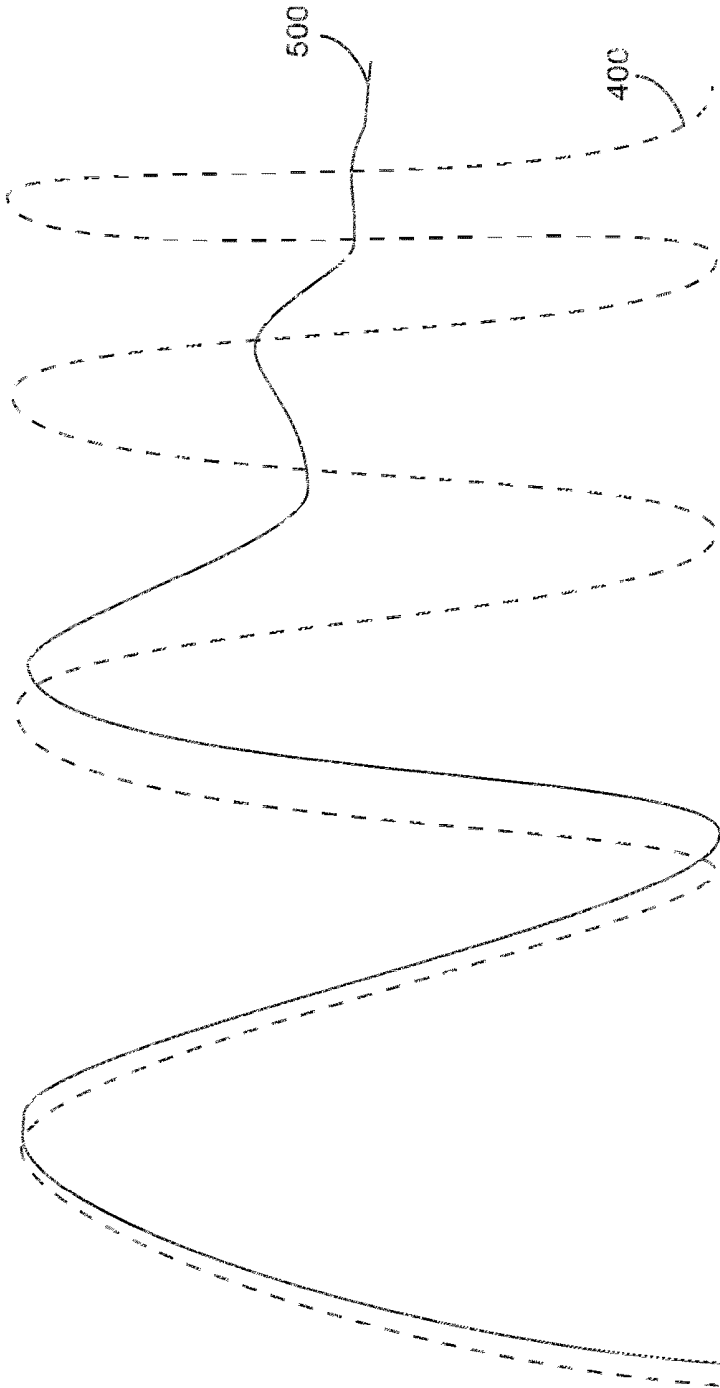


FIG. 5

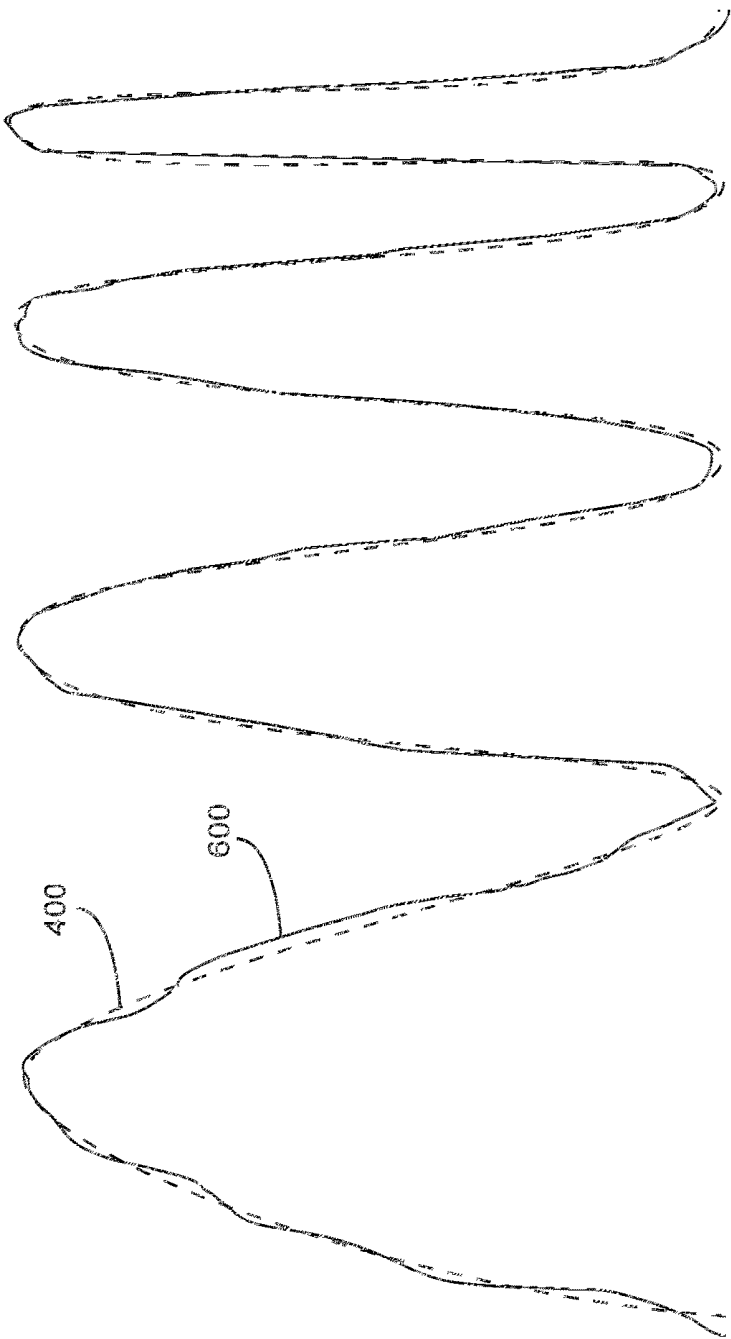


FIG. 6



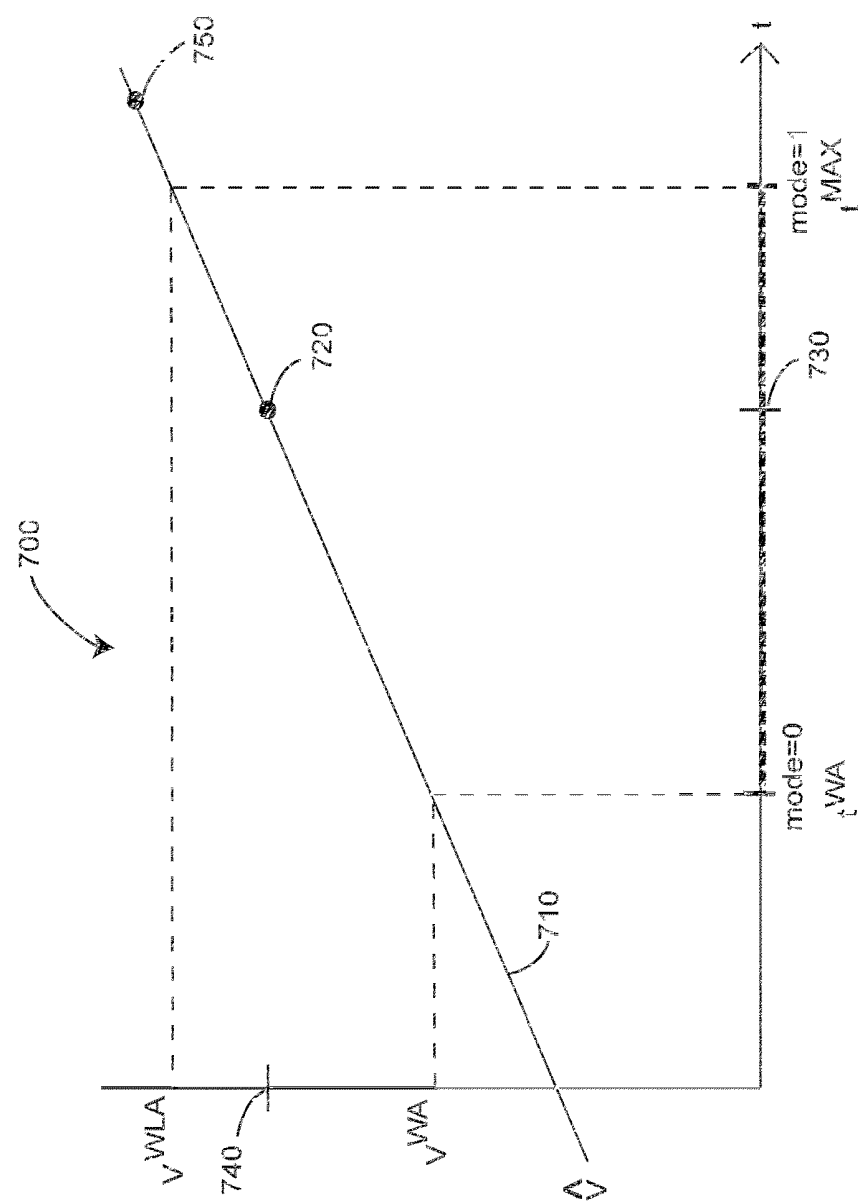


FIG. 7

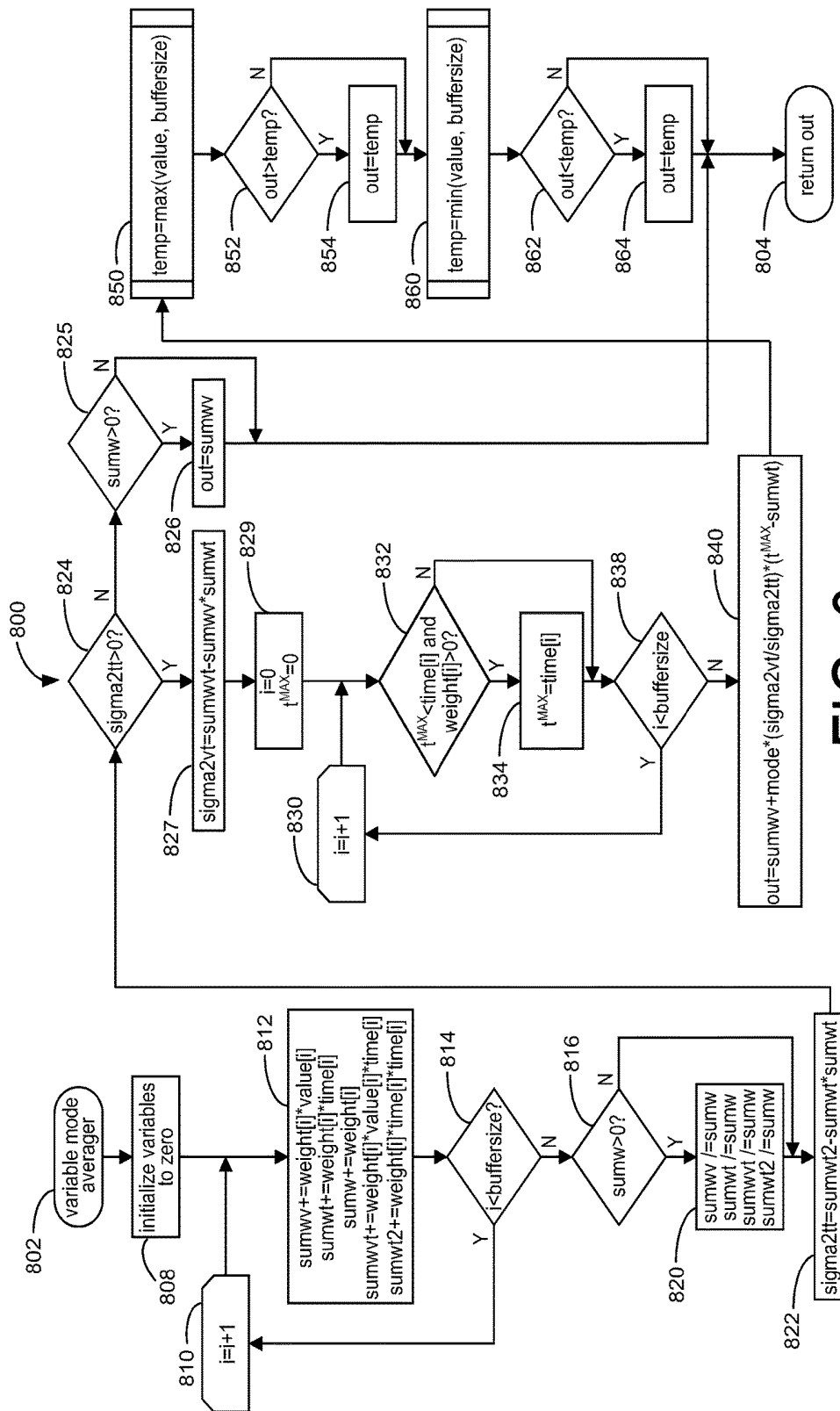


FIG. 8

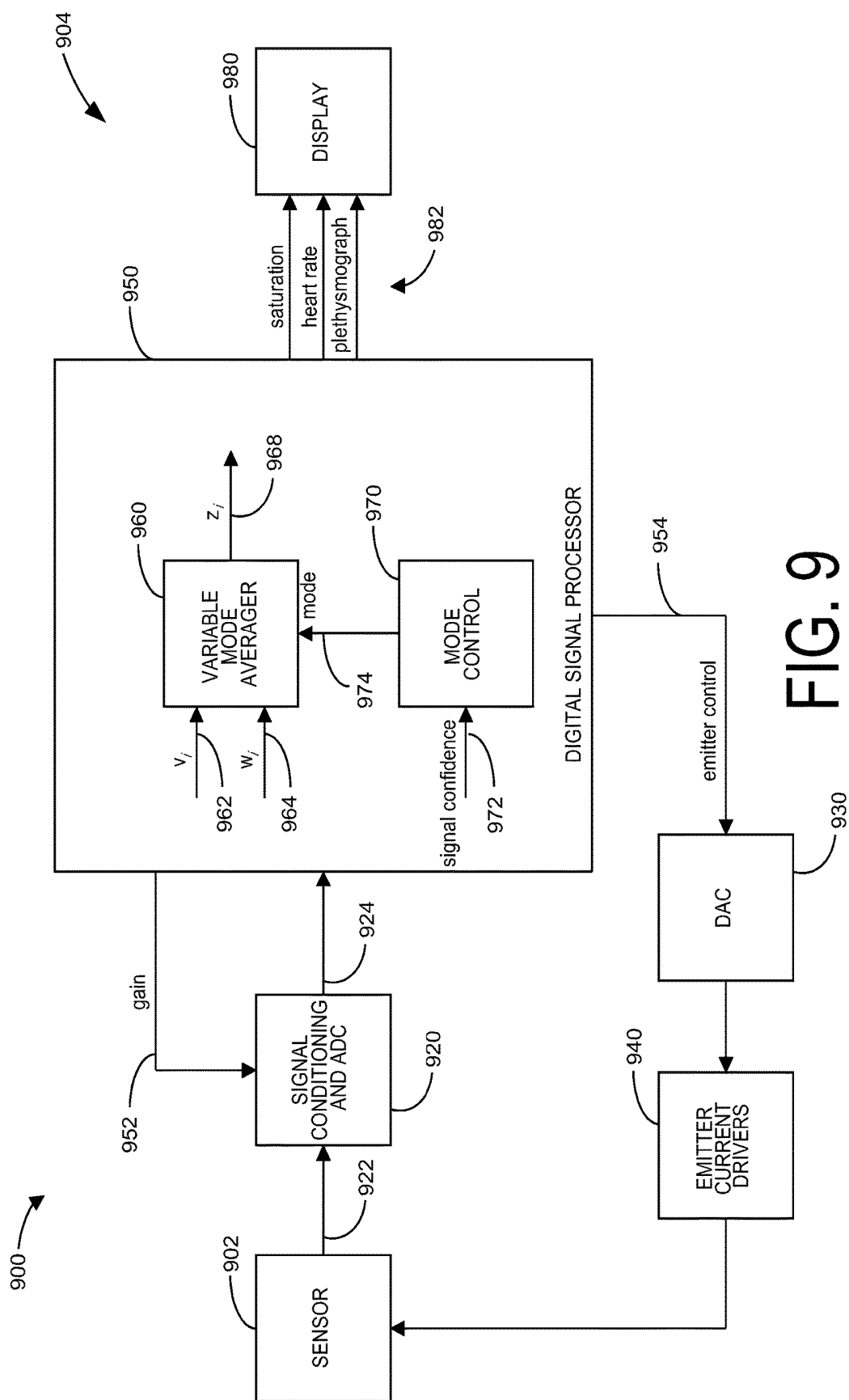


FIG. 9

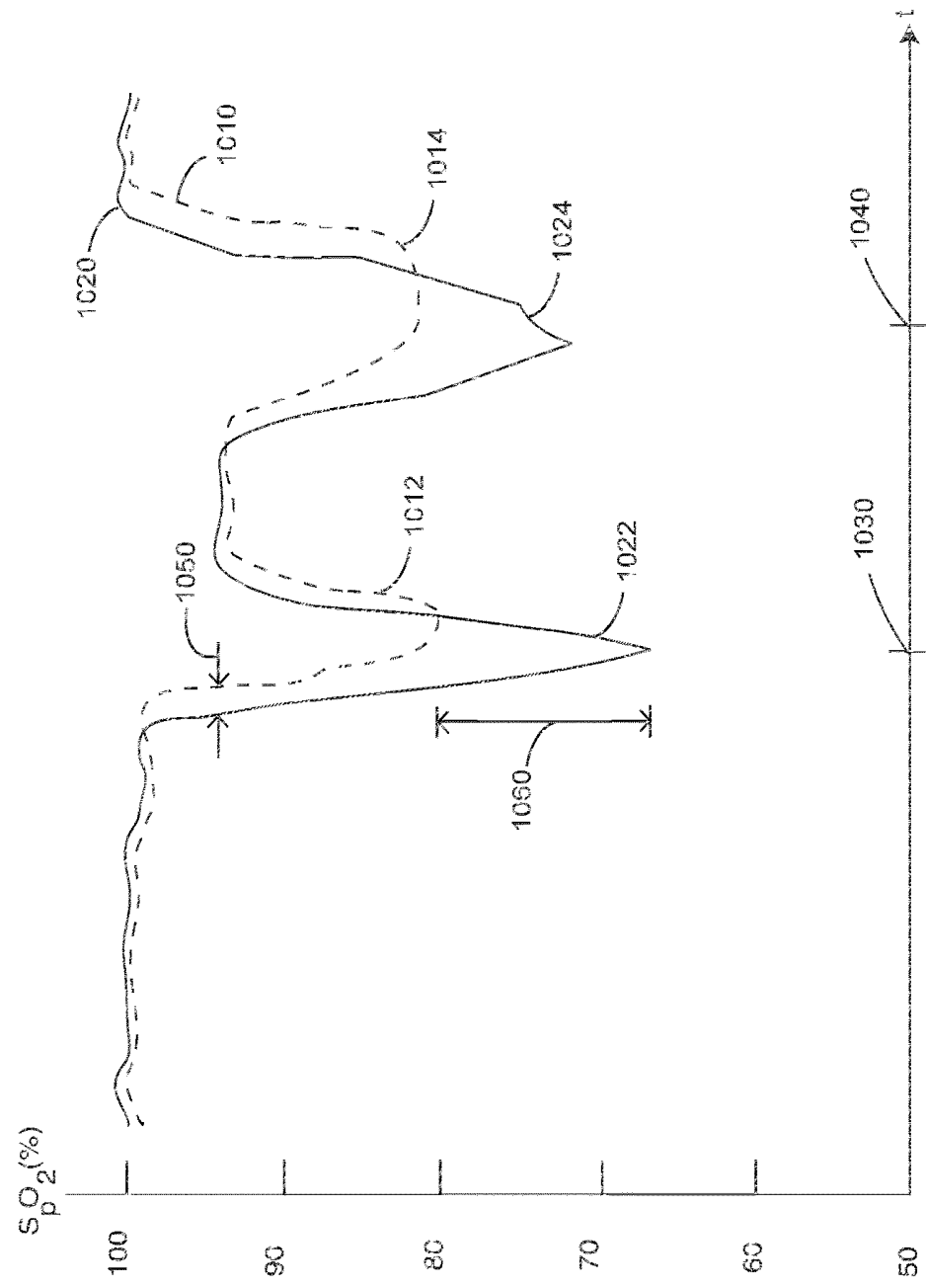


FIG. 10

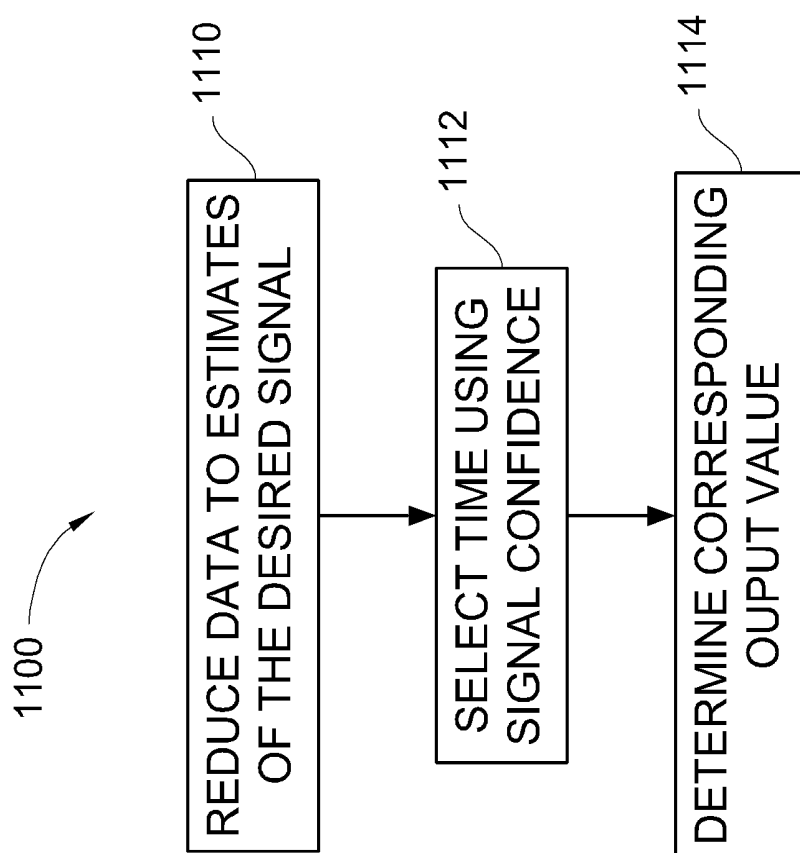


FIG. 11

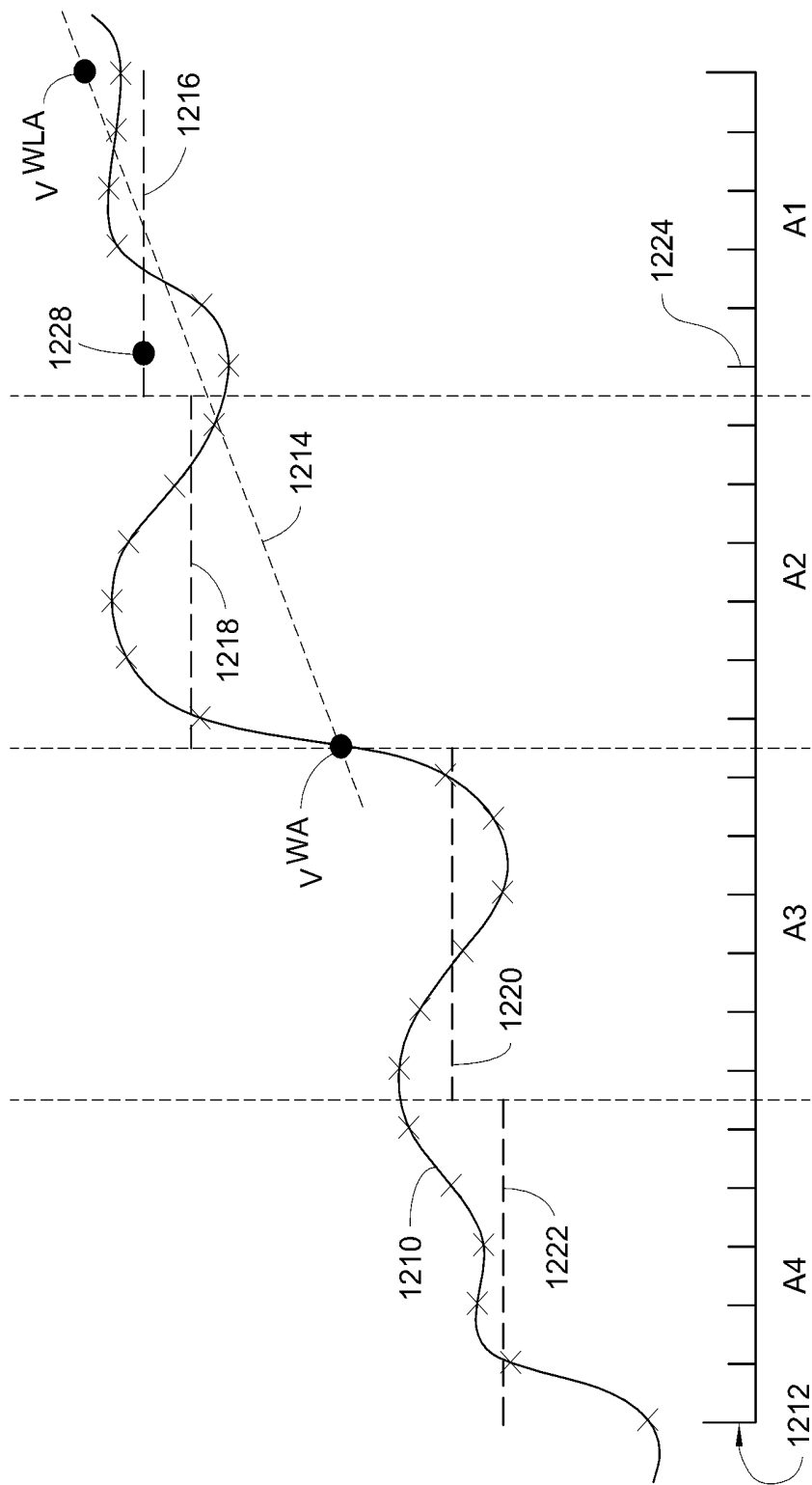


FIG. 12

## VARIABLE INDICATION ESTIMATOR

INCORPORATION BY REFERENCE TO ANY  
PRIORITY APPLICATIONS

[0001] Any and all applications, if any, for which a foreign or domestic priority claim can be identified in the Application Data Sheet of the present application is hereby incorporated by reference under 37 CFR 1.57.

## FIELD OF THE INVENTION

[0002] The present invention is directed to the field of signal processing, and, more particularly, is directed to systems and methods for determining a representative estimate output value for a window of input data.

## BACKGROUND OF THE INVENTION

[0003] Digital signal processing techniques are frequently employed to enhance a desired signal in a wide variety of applications, such as health care, communications and avionics, to name a few. Signal enhancement includes smoothing, filtering and prediction. These processing techniques each operate on a block of input signal values, such as, for example, a window of input signal values, in order to estimate the signal at a specific point in time. FIG. 1 illustrates that smoothing, filtering and prediction can be distinguished by the time at which an output value is generated relative to input values. Shown in FIG. 1 is a time axis 100 and a block 101 of input signal values depicted in this example as occurring within a time window between points  $t^{min}$  and  $t^{max}$ . Specifically, the block 101 includes a set of discrete input values  $\{v_i; i=1, 2, \dots, n\}$  occurring at a corresponding set of time points  $\{t_i; i=1, 2, \dots, n\}$ . A smoother operates on the block 101 of input values to estimate the signal at a time point,  $t_s$  102 between  $t^{min}$  and  $t^{max}$ . That is, a smoother generates an output value based upon input values occurring before and after the output value. A filter operates on the block 101 of input values to estimate the signal at a time  $t_f$  104, corresponding to the most recently occurring input value in the block 101. That is, a filter generates a forward filtered output value at the time  $t_f$  based upon input values occurring at, and immediately before, the output value. A filter also operates on the block 101 to estimate the signal at a time  $t_b$  105 at the beginning of the block 101 to generate a backward filtered value. A forward predictor operates on the block of input values 101 to estimate the signal at time  $t_{pf}$  106, which is beyond the most recently occurring value in the block 101. That is, a forward predictor generates a forward predicted output value based upon input values occurring prior to the output value. A backward predictor operates on the block 101 of input values to estimate the signal at time  $t_{pb}$  108, which is before the earliest occurring value in the block 101. That is, a backward predictor generates a backward predicted output value based upon input values occurring after the output value.

## SUMMARY OF THE INVENTION

[0004] A common smoothing technique uses an average to fit a constant,  $v^A$ , to a set of data values,  $\{v_i; i=1, 2, \dots, n\}$ :

$$v^A = \frac{1}{n} \cdot \sum_{i=1}^n v_i \quad (1)$$

[0005] A generalized form of equation (1) is the weighted average

$$v^{WA} = \frac{\sum_{i=1}^n w_i \cdot v_i}{\sum_{i=1}^n w_i} \quad (2)$$

[0006] Here, each value,  $v_i$ , is scaled by a weight,  $w_i$ , before averaging. This allows data values to be emphasized and de-emphasized relative to each other. If the data relates to an input signal, for example, values occurring during periods of low signal confidence can be given a lower weight and values occurring during periods of high signal confidence can be given a higher weight.

[0007] FIG. 2A illustrates the output of a constant mode averager, which utilizes the weighted average of equation (2) to process a discrete input signal,  $\{v_i; i \text{ an integer}\}$  110. The input signal 110 may be, for example, a desired signal corrupted by noise or a signal having superfluous features. The constant mode averager suppresses the noise and unwanted features, as described with respect to FIG. 5, below. A first time-window 132 defines a first set,  $\{v_i; i=1, 2, \dots, n\}$ , of signal values, which are averaged together to produce a first output value,  $z_1$  122. A second time-window 134, shifted from the previous window 132, defines a second set  $\{v_i; i=2, 3, \dots, n+1\}$  of signal values, which are also averaged together to produce a second output value  $z_2$  124. In this manner, a discrete output signal,  $\{z_j; j \text{ an integer}\}$  120 is generated from a moving weighted average of a discrete input signal  $\{v_i; i \text{ an integer}\}$  110, where:

$$z_j = \sum_{i=j}^{n+j-1} w_i v_i / \sum_{i=j}^{n+j-1} w_i \quad (3)$$

[0008] A common filtering technique computes a linear fit to a set of data values,  $\{v_i; i=1, 2, \dots, n\}$ :

$$\hat{v}_i = \alpha t_i + \beta \quad (4)$$

[0009] where  $\alpha$  and  $\beta$  are constants and  $t_i$  is the time of occurrence of the  $i^{th}$  value. FIG. 2B illustrates the output of a linear mode averager, which uses the linear fit of equation (4) to process a discrete input signal,  $\{v_i; i \text{ an integer}\}$  110. The input signal 110 may be, for example, a desired signal with important features corrupted by noise. The linear mode averager reduces the noise but tracks the important features, as described with respect to FIG. 6 below. A first time-window 132 defines a first set,  $\{v_i; i=1, 2, \dots, n\}$ , of signal values. A linear fit to these  $n$  values is a first line 240, and the value along this line at  $\max\{t_1, t_2, \dots, t_n\}$  is equal to

a first output value,  $z_1$  **222**. A second time-window **134** shifted from the previous window **132** defines a second set,  $\{v_i; i=2, 3, \dots, n+1\}$ , of signal values. A linear fit to these  $n$  values is a second line **250**, and the value along this line at  $\max\{t_2, t_3, \dots, t_{n+1}\}$  is equal to a second output value,  $z_2$  **224**. In this manner, a discrete output signal,  $\{z_j; j \text{ an integer}\}$  **220** is generated from a moving linear fit of a discrete input signal  $\{v_i; i \text{ an integer}\}$ , where:

$$z_j = \alpha_j \cdot t_{n+j-1}^{MAX} + \beta_j \quad (5a)$$

$$t_{n+j-1}^{MAX} = \max\{t_j, t_{j+1}, L_n, t_{n+j-1}\} \quad (5b)$$

**[0010]** In general, the time windows shown in FIGS. 2A-2B may be shifted from each other by more than one input value, and values within each time window may be skipped, i.e., not included in the average. Further, the  $U_s$  may not be in increasing or decreasing order or uniformly distributed, and successive time windows may be of different sizes. Also, although the discussion herein refers to signal values as the dependent variable and to time as the independent variable to facilitate disclosure of the present invention, the concepts involved are equally applicable where the variables are other than signal values and time. For example, an independent variable could be a spatial dimension and a dependent variable could be an image value.

**[0011]** The linear mode averager described with respect to FIG. 2B can utilize a “best” linear fit to the input signal, calculated by minimizing the mean-squared error between the linear fit and the input signal. A weighted mean-squared error can be described utilizing equation (4) as:

$$\mathcal{E}(\alpha, \beta) = \sum_{i=1}^n w_i (v_i - \hat{v}_i)^2 / \sum_{i=1}^n w_i \quad (6a)$$

$$\mathcal{E}(\alpha, \beta) = \sum_{i=1}^n w_i [v_i - (\alpha \cdot t_i + \beta)]^2 / \sum_{i=1}^n w_i \quad (6b)$$

**[0012]** Conventionally, the least-mean-squared (LMS) error is calculated by setting the partial derivatives of equation (6b) with respect to  $\alpha$  and  $\beta$  to zero:

$$\frac{\partial}{\partial \alpha} \mathcal{E}(\alpha, \beta) = 0 \quad (7a)$$

$$\frac{\partial}{\partial \beta} \mathcal{E}(\alpha, \beta) = 0 \quad (7b)$$

**[0013]** Substituting equation (6b) into equation (7b) and taking the derivative yields:

$$-2 \sum_{i=1}^n w_i [v_i - (\alpha \cdot t_i + \beta)] / \sum_{i=1}^n w_i = 0 \quad (8)$$

**[0014]** Solving equation (8) for  $\beta$  and substituting the expression of equation (2) yields:

$$\beta = \frac{\sum_{i=1}^n w_i \cdot v_i}{\sum_{i=1}^n w_i} - \alpha \frac{\sum_{i=1}^n w_i \cdot t_i}{\sum_{i=1}^n w_i} \quad (9a)$$

$$\beta = v^{WA} - \alpha \cdot t^{WA} \quad (9b)$$

**[0015]** where the weighted average time,  $t^{WA}$ , is defined as:

$$t^{WA} = \frac{\sum_{i=1}^n w_i \cdot t_i}{\sum_{i=1}^n w_i} \quad (10)$$

**[0016]** Substituting equation (9b) into equation (4) gives:

$$\hat{v}_i = \alpha(t_i - t^{WA}) + v^{WA} \quad (11)$$

**[0017]** Substituting equation (11) into equation (6a) and rearranging terms results in:

$$\mathcal{E}(\alpha, \beta) = \sum_{i=1}^n w_i [(v_i - v^{WA}) - \alpha \cdot (t_i - t^{WA})]^2 / \sum_{i=1}^n w_i \quad (12)$$

**[0018]** Changing variables in equation (12) gives:

$$\mathcal{E}(\alpha, \beta) = \sum_{i=1}^n w_i (v'_i - \alpha \cdot t'_i)^2 / \sum_{i=1}^n w_i \quad (13)$$

**[0019]** where:

$$v'_i = v_i - v^{WA} \quad (14a)$$

$$t'_i = t_i - t^{WA} \quad (14b)$$

**[0020]** Substituting equation (13) into equation (7a) and taking the derivative yields

$$-2 \sum_{i=1}^n w_i t'_i (v'_i - \alpha \cdot t'_i) / \sum_{i=1}^n w_i = 0 \quad (15)$$

**[0021]** Solving equation (15) for  $\alpha$  gives:

$$\alpha = \frac{\sum_{i=1}^n w_i v'_i t'_i / \sum_{i=1}^n w_i}{\sum_{i=1}^n w_i t'^2_i / \sum_{i=1}^n w_i} \quad (16)$$



[0022] Substituting equations (14a, b) into equation (16) results in:

$$\alpha = \frac{\sum_{i=1}^n w_i (v_i - v^{WA})(t_i - t^{WA}) / \sum_{i=1}^n w_i}{\sum_{i=1}^n w_i (t_i - t^{WA})^2 / \sum_{i=1}^n w_i} \quad (17a)$$

$$\alpha = \frac{\sigma_v^2}{\sigma_t^2} \quad (17b)$$

where:

$$\sigma_v^2 = \sum_{i=1}^n w_i (v_i - v^{WA})(t_i - t^{WA}) / \sum_{i=1}^n w_i \quad (18a)$$

$$\sigma_t^2 = \sum_{i=1}^n w_i (t_i - t^{WA})^2 / \sum_{i=1}^n w_i \quad (18b)$$

[0023] Finally, substituting equation (17b) into equation (11) provides the equation for the least-mean-square (LMS) linear fit to  $\{v_i; i=1, 2, \dots, n\}$ :

$$\hat{v}_i = \frac{\sigma_v^2}{\sigma_t^2} (t_i - t^{WA}) + v^{WA} \quad (19)$$

[0024] FIG. 3 provides one comparison between the constant mode averager, described above with respect to FIG. 2A and equation (2), and the linear mode averager, described above with respect to FIG. 2B and equation (19). Shown in FIG. 3 are input signal values  $\{v_i; i=1, 2, \dots, n\}$  310. The constant mode averager calculates a constant 320 for these values 310, which is equal to  $v^{WA}$ , the weighted average of the input values  $v_i$ . Thus, the constant mode averager output 340 has a value  $v^{WA}$ . For comparison to the linear mode averager, the constant mode averager output can be conceptualized as an estimate of the input values 310 along a linear fit 350, evaluated at time  $t^{WA}$ . The linear mode averager may be thought of as calculating a LMS linear fit,  $\hat{v}_i$  330 to the input signal values,  $v_i$  310. The linear mode averager output 350 has a value,  $v^{WLA}$ . The linear mode averager output is an estimate of the input values 310 along the linear fit 330, described by equation (19), evaluated at an index  $i$  such that  $t_i = t^{MAX}$ :

$$v^{WLA} = \frac{\sigma_v^2}{\sigma_t^2} (t^{MAX} - t^{WA}) + v^{WA} \quad (20)$$

[0025] where:

$$t^{MAX} = \max\{t_1, t_2, \dots, t_n\} \quad (21)$$

[0026] As illustrated by FIG. 3, unlike the constant mode averager, the linear mode averager is sensitive to the input signal trend. That is, the constant mode averager provides a constant fit to the input values, whereas the linear mode averager provides a linear fit to the input values that corresponds to the input value trend. As a result, the output of the linear mode averager output responds faster to changes in

the input signal than does the output of the constant mode averager. The time lag or delay between the output of the constant mode averager and the output of the linear mode averager can be visualized by comparing the time difference 360 between the constant mode averager output value 340 and the linear mode averager output value 350.

[0027] FIGS. 4-6 illustrate further comparisons between the constant mode averager and the linear mode averager. FIG. 4 depicts a noise-corrupted input signal 410, which increases in frequency with time. FIGS. 5-6 depict the corresponding noise-free signal 400. FIG. 5 also depicts the constant mode averager output 500 in response to the input signal 410, with the noise-free signal 400 shown for reference. FIG. 6 depicts the linear mode averager output 600 in response to the input signal 410, with the noise-free signal 400 also shown for reference. As shown in FIG. 5, the constant mode averager output 500 suppresses noise from the input signal 410 (FIG. 4) but displays increasing time lag and amplitude deviation from the input signal 400 as frequency increases. As shown in FIG. 6, the linear mode averager output 600 tends to track the input signal 400 but also tracks a portion of the noise on the input signal 410.

[0028] FIGS. 4-6 suggest that it would be advantageous to have an averager that has variable characteristics between those of the linear mode averager and those of the constant mode averager, depending on signal confidence. Specifically, it would be advantageous to have a variable mode averager that can be adjusted to track input signal features with a minimal output time lag when signal confidence is high and yet adjusted to smooth an input signal when signal confidence is low. Further, it would be advantageous to have a variable mode averager that can be adjusted so as not to track superfluous input signal features regardless of signal confidence.

[0029] One aspect of the present invention is a variable mode averager having a buffer that stores weighted input values. A mode input specifies a time value relative to the input values. A processor is coupled to the buffer, and the processor is configured to provide an estimate of the input values that corresponds to the time value. In a particular embodiment, the mode input is adjustable so that the estimate varies between that of a smoother and that of a forward predictor of the input values. In another embodiment, the mode input is adjustable so that the estimate varies between that of a smoother and that of a filter of the input values. In yet another embodiment, the mode input is adjustable so that the estimate varies between that of an average of the input values and that of a filter of the input values. The mode input may be adjustable based upon a characteristic associated with the input values, such as a confidence level. In one variation of that embodiment, the estimate can be that of a smoother when the confidence level is low and that of a filter when the confidence level is high. The estimate may occur along a curve-fit of the input values at the time value. In one embodiment, the curve-fit is a linear LMS fit to the input values.

[0030] Another aspect of the present invention is a signal averaging method. The method includes identifying signal values and determining weights corresponding to the signal values. The method also includes computing a trend of the signal values adjusted by the weights. Further, the method includes specifying a time value relative to the signal values based upon a characteristic associated with the signal values and estimating the signal values based upon the trend

evaluated at the time value. The method may also incorporate the steps of determining a confidence level associated with the signal values and specifying the time value based upon the confidence level. In one embodiment, the trend is a linear LMS fit to the signal values adjusted by the weights. In that case, the time value may generally correspond to the maximum time of the signal values when the confidence level is high and generally correspond to the weighted average time of the signal values when the confidence level is low.

[0031] Yet another aspect of the present invention is a signal averaging method having the steps of providing an input signal, setting a mode between a first mode value and a second mode value and generating an output signal from an estimate of the input signal as a function of said mode. The output signal generally smoothes the input signal when the mode is proximate the first mode value, and the output signal generally tracks the input signal when the mode is proximate the second mode value. The method may also include determining a characteristic of the input signal, where the setting step is a function of the characteristic. In one embodiment, the characteristic is a confidence level relating to the input signal. In another embodiment, the setting step incorporates the substeps of setting the mode proximate the first mode value when the confidence level is low and setting the mode proximate the second mode value when the confidence level is high. In another embodiment, the input signal is a physiological measurement and the setting step comprises setting the mode proximate the first mode value when the measurement is corrupted with noise or signal artifacts and otherwise setting the mode proximate the second mode value so that the output signal has a fast response to physiological events.

[0032] A further aspect of the present invention is a signal averager having an input means for storing signal values, an adjustment means for modifying the signal values with corresponding weights, a curve fitting means for determining a trend of the signal values, and an estimate means for generating an output value along the trend. The signal averager may further have a mode means coupled to the estimate means for variably determining a time value at which to generate the output value.

[0033] For purposes of summarizing the invention, certain aspects, advantages and novel features of the invention have been described herein. Of course, it is to be understood that not necessarily all such aspects, advantages or features will be embodied in any particular embodiment of the invention.

#### BRIEF DESCRIPTION OF THE DRAWINGS

[0034] A general architecture that implements the various features of the invention will now be described with reference to the drawings. The drawings and the associated descriptions are provided to illustrate embodiments of the invention and not to limit the scope of the invention. Throughout the drawings, reference numbers are re-used to indicate correspondence between referenced elements. In addition, the first digit of each reference number indicates the figure in which the element first appears.

[0035] FIG. 1 is a time graph depicting the output of conventional smoother, filter and predictor signal processors;

[0036] FIG. 2A is an amplitude versus time graph depicting the output of a conventional constant mode averager;

[0037] FIG. 2B is an amplitude versus time graph depicting the output of a conventional linear mode averager;

[0038] FIG. 3 is an amplitude versus time graph comparing the outputs of a constant mode averager and a linear mode averager;

[0039] FIG. 4 is an amplitude versus time graph depicting a noisy input signal;

[0040] FIG. 5 is an amplitude versus time graph depicting a constant mode averager output signal corresponding to the input signal of FIG. 4;

[0041] FIG. 6 is an amplitude versus time graph depicting a linear mode averager output signal corresponding to the input signal of FIG. 4;

[0042] FIG. 7 is an amplitude versus time graph illustrating the characteristics of one embodiment of the variable mode averager;

[0043] FIG. 8 is a flow chart of a variable mode averager embodiment;

[0044] FIG. 9 is a block diagram illustrating a variable mode averager applied to a pulse oximeter; and

[0045] FIG. 10 is an oxygen saturation output versus time graph for a pulse oximeter utilizing a variable mode averager.

[0046] FIG. 11 is a flow chart of an output value selection process of a signal processor, according to an embodiment of the invention.

[0047] FIG. 12 is an amplitude versus time graph depicting exemplary potential output values of the output value selection process of FIG. 11, according to an embodiment of the invention.

#### DETAILED DESCRIPTION OF THE PREFERRED EMBODIMENTS

[0048] FIG. 7 illustrates the output characteristics of a variable mode averager according to the present invention. The output of the variable mode averager is a mode-dependent weighted linear average (MWLA) defined as

$$v^{MWLA} = \text{mode} \cdot \frac{\sigma_w^2}{\sigma_n^2} (t^{MAX} - t^{WA}) + v^{WA} \quad (22)$$

[0049] Equation (22) is a modified form of equation (20), which is motivated by equations (2) and (19) along with recognition of the relationships in Table 1.

TABLE 1

| VARIABLE MODE AVERAGER OUTPUT |                        |                      |                        |
|-------------------------------|------------------------|----------------------|------------------------|
|                               | mode = 0               | mode = 1             | any mode $\exists$ 0   |
| Processing Function           | Constant Mode Averager | Linear Mode Averager | Variable Mode Averager |
| Output                        | $v^{WA}$               | $v^{WLA}$            | $v^{MWLA}$             |
| Defining Formula              | Equation (2)           | Equation (20)        | Equation (22)          |
| Processing Method             | Weighted Average       | LMS Linear Fit       | FIG. 8                 |

[0050] As shown in Table 1, the Variable Mode Averager in accordance with the present invention includes the constant mode averager processing function and the linear mode averager processing function, which are known processing functions. As further shown in Table 1, the Variable Mode

Averager of the present invention also includes a variable mode averager processing function, which will be described below.

**[0051]** As shown in Table 1, if mode=0, the variable mode averager output is  $v^{WA}$ , the output of the constant mode averager function, which utilizes a weighted average of the input signal values. If mode=1, the variable mode averager output is  $v^{WLA}$ , the output of the linear mode averager function, which utilizes a LMS linear fit to the input signal values. If  $0 < \text{mode} < 1$ , then the variable mode averager output is  $v^{MWLA}$  and has output characteristics that are between that of the constant mode averager and the linear mode averager. In addition, if mode>1, then the variable mode averager behaves as a forward predictor.

**[0052]** As shown in FIG. 7, the variable mode averager output 720 is an estimate of the input values at a selected time along the linear fit 710, which indicates a trend of the input values. Assuming  $0 < \text{mode} < 1$ , the mode variable determines the equivalent time 730 between  $t^{WA}$  and  $t^{MAX}$  for which the estimate is evaluated, yielding an output value 740 between  $v^{WA}$  and  $v^{WLA}$ . Thus, the mode variable acts to parametrically vary the time delay between the input and output signals of the variable mode averager, along with associated output characteristics. If mode=0, the time delay 360 (FIG. 3) is that of the constant mode averager. If mode=1, there is no time delay. If mode>1, the variable mode averager is predicting a future input value based on n past values. In this manner, the variable mode averager can be used to advantageously adjust between the smoothing characteristics of the constant mode averager and the tracking characteristics of the linear mode averager, as described above with respect to FIGS. 4-6. The variable mode control determines how much of each particular characteristic to use for a particular input signal and application. For example, for time periods when the input signal has low confidence, mode can be set further towards zero, although with a time lag penalty. For time periods when the input signal has high confidence or when minimum time lag is required, mode can be set further towards one, or even to a value greater than one.

**[0053]** The variable mode averager has been described in terms of weighted input values. One of ordinary skill, however, will recognize that the present invention includes the case where all of the weights are the same, i.e., where the input values are equally weighted or unweighted. Further, although the variable mode averager has been described in terms of a linear mode averager, one of ordinary skill in the art will recognize that a variable mode averager could also be based on non-linear curve fits, such as exponential or quadratic curves indicating a non-linear trend of the input signal. In addition, one of ordinary skill will understand that the variable mode averager can be implemented to operate on continuous data as well as infinitely long data. Also, a variable mode averager based upon a linear fit by some criteria other than LMS; a variable mode averager using any mode value, including negative values; and a variable mode averager based upon a linear fit where  $t^{min} = \min\{t_1, t_2, \dots, t_n\}$  is substituted for  $t^{MAX}$  in equation (22) are all contemplated as within the scope of the present invention.

**[0054]** FIG. 8 illustrates one embodiment 800 of a variable mode signal averager. After an entry point 802, variables are initialized to zero in a block 808. Next, in a block 812, the sums of various parameters are calculated by summing the products of corresponding values in each of three buffers: an

input data buffer, value[i]; a weight buffer, weight[i]; and a time value buffer, time[i]. In addition, the weight[i] values are summed. These sums are calculated over the entire length of each buffer, representing a single time window of n values. The calculations are performed by incrementing a loop counter i in a block 810 and reentering the block 812. The loop counter i specifies a particular value in each buffer. Each time through the block 812, the variable mode signal averager generates products of buffer values and adds the results to partial sums. After completing the partial sums, the variable mode signal averager then determines if the ends of the buffers have been reached in a decision block 814 by comparing the incremented value of i to the size of the buffer. If the ends of the buffers have not been reached, the variable mode averager increments the loop counter i and reenters the block 812; otherwise, the variable mode averager continues to a decision block 816.

**[0055]** In the decision block 816, a check is made whether the sum of the weights, sumw, is greater than zero. If so, each of the sums of the products from the block 812 is divided by sumw in a block 820. In the block 820, the parameters computed are:

**[0056]** sumwv, the weighted average value of equation (2);

**[0057]** sumwt, the weighted average time of equation (10);

**[0058]** sumwvt, the weighted average product of value and time; and

**[0059]** sumwt2, the weighted average product of time squared.

**[0060]** The sumwt2 parameter from the block 820 is then used in a block 822 to calculate an autocovariance  $\sigma_{2t}$  in accordance with equation (18b). If, in a decision block 824, a determination is made that the autocovariance is not greater than zero, then in a decision block 825, a determination is made whether the sum of the weights is greater than zero. If, in the decision block 825, the sum of the weights is not greater than zero, then an output value, out, which was initialized to zero in the block 808, is returned as a zero value at a termination point 804. Otherwise, if, in the decision block 825, a determination is made that the sum of the weights is greater than zero, then in a block 826, the value of the sum of the weights is assigned to the output value, out, and the output value is then returned at the termination point 804.

**[0061]** If, in the decision block 824, the autocovariance is determined to be greater than zero, then in a block 827, the sumwvt parameter from the block 820 is used to calculate a crossvariance signal  $\sigma_{2vt}$  in accordance with equation (18a). Thereafter, the maximum time,  $t^{MAX}$ , as defined in equation (21), is determined by finding the largest time value in the time buffer, time[i]. In particular, in a block 829, the loop counter, i, is reinitialized to zero and the value of  $t^{MAX}$  is initialized to zero. Next, in a decision block 832, the current value of  $t^{MAX}$  is compared to the current value of the time buffer indexed by the loop counter, i. If the current value of  $t^{MAX}$  is not less than the current value of the time buffer or if the current weight value indexed by i is not greater than zero, then  $t^{MAX}$  is not changed and a block 834 is bypassed. On the other hand, if the current value of  $t^{MAX}$  is less than the current time value and if the current weight value is greater than zero, then the block 834 is entered, and the value of  $t^{MAX}$  is replaced with the current time value time[i]. In either case, in a decision block 838, the loop counter, i, is compared to the buffer size, and, if the loop

counter,  $i$ , is less than the buffer size, the loop counter,  $i$ , is incremented in a block **830**, and the comparisons are again made in the decision block **832**.

[0062] When, in the decision block **838**, it is determined that the loop counter,  $i$ , has reached the buffer size, the variable mode averager proceeds to a block **840** with the largest value of  $\text{time}[i]$  saved as the value of  $t^{\text{MAX}}$ . In the block **840**, a single output value,  $\text{out}$ , is computed in accordance with equation (22). Thereafter, the output value,  $\text{out}$ , is limited to the range of values in the input data buffer,  $\text{value}[i]$ . This is accomplished by comparing  $\text{out}$  to the maximum and minimum values in the data buffer. First, in a block **850**, the maximum of the value buffer is determined. Then, in a decision block **852**, the maximum of the value buffer is compared to  $\text{out}$ . If  $\text{out}$  is bigger than the maximum of the value buffer, then, in a block **854**,  $\text{out}$  is limited to the maximum value in the buffer. Otherwise, the block **854** is bypassed, and  $\text{out}$  remains as previously calculated in the block **840**. Thereafter, in a block **860**, the minimum of the value buffer is determined. The minimum of the value buffer is compared to  $\text{out}$  in a decision block **862**. If  $\text{out}$  is smaller than the minimum of the value buffer, then, in a block **864**,  $\text{out}$  is set to the minimum value in the buffer. Otherwise, the block **864** is bypassed, and  $\text{out}$  is not changed. The value of  $\text{out}$  determined by the block **840**, the block **852** or the block **864** is then returned from the routine via the termination point **804**.

[0063] In one embodiment, the process described with respect to FIG. **8** is implemented as firmware executing on a digital signal processor. One of ordinary skill in the art will recognize that the variable mode averager can also be implemented as a digital circuit. Further, a variable mode averager implemented as an analog circuit with analog inputs and outputs is also contemplated to be within the scope of the present invention.

[0064] Pulse oximetry is one application that can effectively use signal processing techniques to provide caregivers with improved physiological measurements. Pulse oximetry is a widely accepted noninvasive procedure for measuring the oxygen saturation level of arterial blood, an indicator of oxygen supply. Early detection of low blood oxygen is critical in the medical field, for example in critical care and surgical applications, because an insufficient supply of oxygen can result in brain damage and death in a matter of minutes. Pulse oximeter systems are described in detail in U.S. Pat. No. 5,632,272, U.S. Pat. No. 5,769,785, and U.S. Pat. No. 6,002,952, which are assigned to the assignee of the present invention and which are incorporated by reference herein.

[0065] FIG. **9** depicts a general block diagram of a pulse oximetry system **900** utilizing a variable mode averager **960**. A pulse oximetry system **900** consists of a sensor **902** attached to a patient and a monitor **904** that outputs desired parameters **982** to a display **980**, including blood oxygen saturation, heart rate and a plethysmographic waveform. Conventionally, a pulse oximetry sensor **902** has both red (RED) and infrared (IR) light-emitting diode (LED) emitters (not shown) and a photodiode detector (not shown). The sensor **902** is typically attached to a patient's finger or toe, or to a very young patient's foot. For a finger, the sensor **902** is configured so that the emitters project light through the fingernail and into the blood vessels and capillaries underneath. The photodiode is positioned at the fingertip opposite the fingernail so as to detect the LED transmitted light as it

emerges from the finger tissues, producing a sensor output **922** that indicates arterial blood absorption of the red and infrared LED wavelengths.

[0066] As shown in FIG. **9**, the sensor output **922** is coupled to analog signal conditioning and an analog-to-digital conversion (ADC) circuit **920**. The signal conditioning filters and amplifies the analog sensor output **922**, and the ADC provides discrete signal values to the digital signal processor **950**. The signal processor **950** provides a gain control **952** to amplifiers in the signal conditioning circuit **920**. The signal processor **950** also provides an emitter control **954** to a digital-to-analog conversion (DAC) circuit **930**. The DAC **930** provides control signals for the emitter current drivers **940**. The emitter drivers **940** couple to the red and infrared LEDs in the sensor **902**. In this manner, the signal processor **950** can alternately activate the sensor LED emitters and read the resulting output **922** generated by the photodiode detector.

[0067] The digital signal processor **950** determines oxygen saturation by computing the differential absorption by arterial blood of the red and infrared wavelengths emitted by the sensor **902**. Specifically, the ADC **920** provides the processor **950** with a digitized input **924** derived from the sensor output **922**. Based on this input **924**, the processor **950** calculates ratios of detected red and infrared intensities. Oxygen saturation values,  $v_r$ , are empirically determined based on the calculated red and infrared ratios. These values are an input signal **962** to the variable mode averager **960**. Each of the input values, are associated with weights,  $w_r$ , which form a second input **964** to the averager **960**. The individual weights,  $w_r$ , are indicative of the confidence in particular ones of the corresponding saturation values,  $v_r$ . A third input **974** sets the mode of the averager **960**. The variable mode averager **960** processes the values,  $v_r$ , weights,  $w_r$ , and mode as described above with respect to FIGS. **7-8** to generate values,  $z_r$ . The values  $z_r$  are the averager output **968**, from which is derived the saturation output **982** to the display **980**.

[0068] The mode signal may be generated by an external source (not shown) or it may be generated by another function within the digital signal processor. For example, mode may be generated from the confidence level of the input signal as illustrated in FIG. **9**. FIG. **9** illustrates a signal confidence input **972** to a mode control process **970**. The mode control process **970** maps the signal confidence input **972** to the mode input **974** of the variable mode averager **960**. When the signal confidence is low, the mode control **970** sets mode to a relatively small value. Depending on the application, mode may be set close to zero. When the signal confidence is high, the mode control **970** sets mode to a relatively large value. Some applications may prefer a mode of one for a high signal confidence, but this is not a requirement. When the signal confidence is neither high nor low, mode is set to an intermediate value (in some applications, mode may be set to a value between zero and one) empirically to achieve a reasonable tradeoff between a fast saturation output response and saturation accuracy.

[0069] The signal quality of pulse oximetry measurements is adversely affected by patients with low perfusion of blood, causing a relatively small detected signal, ambient noise, and artifacts caused by patient motion. The signal confidence input **972** is an indication of the useful range of the pulse oximetry algorithms used by the digital signal processor **950** as a function of signal quality. This useful range is

extended by signal extraction techniques that reduce the effects of patient motion, as described in U.S. Pat. No. 5,632,272, U.S. Pat. No. 5,769,785, and U.S. Pat. No. 6,002,952, referenced above. Signal confidence is a function of how well the sensor signal matches pulse oximetry algorithm signal models. For example, the red and infrared signals should be highly correlated and the pulse shapes in the pulsatile red and infrared signals should conform to the shape of physiological pulses, as described in U.S. patent application Ser. No. 09/471,510 filed Dec. 23, 1999, entitled Plethysmograph Pulse Recognition Processor, which is assigned to the assignee of the present invention and which is incorporated by reference herein. As a particular example, signal confidence can be determined by measuring pulse rate and signal strength. If the measured signal strength is within an expected range for the measured pulse rate, then the confidence level will be high. On the other hand, if the measured signal strength is outside the expected range (e.g., too high for the measured pulse rate), then the confidence level will be low. Other measured or calculated parameters can be advantageously used to set the confidence level.

[0070] FIG. 10 illustrates the oxygen saturation output of a pulse oximeter utilizing a variable mode averager, as described above with respect to FIG. 9. A first output 1010 illustrates oxygen saturation versus time for input oxygen saturation values processed by a conventional weighted averager or, equivalently, by a variable mode averager 960 with  $\text{mode} \approx 0$ . A second output 1020 illustrates oxygen saturation versus time for the variable mode averager 960 with  $\text{mode} \approx 1$ . Each output 1010, 1020 indicates exemplary desaturation events occurring around a first time 1030 and a second time 1040. The desaturation events correspond to a patient experiencing a potentially critical oxygen supply shortage due to a myriad of possible physiological problems. With  $\text{mode} \approx 1$ , the variable mode averager responds to the onset of the desaturation events with less lag time 1050 than that of the conventional weighted average. Further, the variable mode averager responds to the full extent of the desaturations 1060 whereas the conventional weighted average does not. When signal confidence is low, the variable mode averager is adjusted to provide similar smoothing features to those of a conventional weighted average. When signal confidence is high, however, the variable mode averager is advantageously adjusted to respond faster and more accurately to a critical physiological event. The fast response advantage of the variable mode averager has other physiological measurement applications, such as blood-pressure monitoring and ECG.

[0071] The variable mode averager has been disclosed in detail in connection with various embodiments of the present invention. One of ordinary skill in the art will appreciate many variations and modifications within the scope of this invention.

[0072] Thus, the variable mode averager disclosed in the foregoing advantageously allows a signal processor the ability to reduce a window of input values of, for example, a noisy signal, to a linear fit of estimates of the desired signal, where a selected output value from the estimates corresponds at least in part to the selection of a time or mode. The mode can correspond, for example, to a degree of confidence that the most recently received input signal is an accurate representation of the desired signal. However, a skilled artisan will recognize from the disclosure herein that

other mechanisms can be used to reduce a set of input values to one or more appropriate output values.

[0073] For example, FIG. 11 illustrates a flow chart of an output value selection process 1100 of a signal processor, according to an embodiment of the invention. As shown in FIG. 11, the process 1100 includes BLOCK 1110, where the signal processor reduces a set or window of input values to one or more or a set of estimates such as the foregoing linear fit of the variable mode averager, or the like. The process 1100 then moves to BLOCK 1112, where the processor selects a time based, for example, on an indication of confidence that the set of input values represents a desired signal. The process 1100 in BLOCK 1114 then determines the output value from the one or more, or set of estimates, which corresponds to the selected time.

[0074] As will be appreciated by an artisan from the disclosure herein, a wide variety of processes or mechanisms can be used to reduce a set or window of input data to a set of estimates. For example, the processor can execute the foregoing variable mode averager, or other more conventional signal processing techniques, such as, for example, simple averaging, weighted averaging, linear averaging, filtering, prediction, or the like to reduce the set of input data before selecting an appropriate time using the mode or signal confidence.

[0075] According to one embodiment, the processor can reduce input data through segmentation of a window of input values. For example, FIG. 12 illustrates an amplitude versus time graph depicting an input signal 1210, including a window 1212 of input values. According to one embodiment, the input signal 1210 comprises, for example, a desired signal corrupted by noise or a signal having superfluous features. FIG. 12 shows an example of reduction of the input values corresponding to the window 1212 to the linear fit 1214 of estimates using the foregoing variable mode averager. As disclosed in the foregoing, when  $0 < \text{mode} < 1$ , the mode variable determines the equivalent time along the linear fit of estimates for which an output estimate can be evaluated, thereby yielding an output value between  $v^{WA}$  and  $v^{WLA}$ .

[0076] However, FIG. 12 also shows reduction of the input values using segmentation. For example, a signal processor can segment the window 1212 of input values into a plurality of segments, e.g., Segments A1, A2, A3, and A4. A artisan will recognize from the disclosure herein that the use of four segments in FIG. 12 is for illustration only, and the number of segments can be selected based on a number of factors, such as, for example, the number of input values in the window, signal processing speed and capacity, experimental results, or the like.

[0077] According to one embodiment, the signal processor then determines one or more or a set of estimates corresponding to each segment. For example, in a straightforward implementation, the signal processor may select simple weighted averages 1216, 1218, 1220, 1222, as estimates for each of the Segments A1, A2, A3, and A4, respectively, of the window 1212 of input values. However, an artisan will recognize from the disclosure herein that the estimates for each segment may range in complexity from simple selection of one or more of the input values, to more complex calculations, such as application of the foregoing variable mode averager or the like for the input values of each segment. Moreover, the artisan will recognize from the disclosure herein that the signal confidence indicator could

be used to select one, some, or all of the input values corresponding to one, some, or all, of the segments for the generation of the estimate values.

[0078] Once the estimates for each segment are determined, the signal processor selects a time corresponding to a degree of confidence that the input values represent a desired signal. A signal confidence indicator representative of whether the more recently received input signal values are accurate representations of a desired signal can be derived from, for example, an analysis of the amount of noise in the signal, comparing the signal to expected patterns or templates, or the like. The analysis of noise can include a measurement of the entropy of the signal, adherence of the signal to predetermined mathematical models based on a priori information about the expected or desired signal, or the like.

[0079] In the example illustrated in FIG. 12, the signal processor may have higher confidence that the estimates from the segmentation are representative of the desired signal, and therefore choose a time 1224 where the estimates 1216-1222 are to be evaluated. According to an embodiment using a more straightforward reduction of the segments, such as, for example, the simple weighted averaging, the signal processor may interpolate between estimates, such as, output value 1228. When more complex mechanisms are used to reduce the input data, determination of the output value 1228 may be directly calculated, such as, for example, calculation of the output value using the variable mode averager. A skilled artisan will also recognize from the disclosure herein that the output value 1228 may comprise an interpolation between more complex estimates, such as, for example, zero, first, second, etc. order interpolation.

[0080] Selection of the time 1224 allows the signal processor to slide the output value along, for example, the exemplary line 1214 or one of the segment estimates 1216-1222, thereby providing an output value deemed likely to indicate the value of the desired signal for the most recent input value of the time window 1212. For example, as disclosed in the foregoing, when the signal confidence indicator represents a higher confidence in the input values, the output value 1228 may slide toward the most recent input values, whereas the output value 1228 may side in the opposite direction during a time of lower signal confidence.

[0081] The signal processing techniques disclosed in the foregoing, which use a confidence measure to select an output value from a set of estimates of a window of input values, is particular applicable to the monitoring of critical physiological parameters in patient-care settings. When applied to pulse oximeter oxygen saturation measurements, the mode parameter can be varied in real-time to achieve a tradeoff between the suppression of false alarms and signal artifacts and the immediate detection of life threatening oxygen desaturation events. For example, during the monitoring of physiological parameters, it is often common for motion artifacts or other abnormalities to appear in the input value stream. Such abnormalities often decrease the confidence measure, or mode, being used by the signal processor. As disclosed in the foregoing, a lower signal confidence may lead to the signal processor selecting a smoothed output estimate for a specific time window, such as for example, time windows ranging from approximately 15 seconds to over 1 minute, thereby avoiding crossing over alarm-activating output thresholds. Alternatively, as discussed with reference to FIG. 10, a signal abnormality accompanied by

high signal confidence leads the signal processor to the selection of an output estimate that more accurately reflects the extent of a potentially life threatening desaturation event, thereby ensuring an appropriate alarm activation.

[0082] Although the foregoing invention has been described in terms of certain preferred embodiments, other embodiments will be apparent to those of ordinary skill in the art from the disclosure herein. Additionally, other combinations, omissions, substitutions and modifications will be apparent to the skilled artisan in view of the disclosure herein. Accordingly, the present invention is not intended to be limited by the reaction of the preferred embodiments which disclose by way of example only, but is to be defined by reference to the appended claims.

[0083] Additionally, all publications, patents, and patent applications mentioned in this specification are herein incorporated by reference to the same extent as if each individual publication, patent, or patent application was specifically and individually indicated to be incorporated by reference.

#### 1.-18. (canceled)

19. A pulse oximetry system, the pulse oximetry system comprising:

an analog signal conditioning circuit that receives signals corresponding to attenuated red and infrared light transmitted through body tissue of a patient, the analog signal conditioning circuit comprising an analog-to-digital conversion circuit that provides first discrete signal values at a first time and second discrete signal values at a second time;

a digital-to-analog conversion circuit that provides control signals to emitter current drivers to cause the red and infrared light to transmit through the body tissue; and

a digital signal processor that:

receives the first discrete signal values and the second discrete signal values from the analog signal conditioning circuit, the first discrete signal values having a first amount of noise greater than a second amount of noise of the second discrete signal values;

outputs an emitter control signal to the digital-to-analog conversion circuit;

calculates first oxygen saturation values based on the first discrete signal values and second oxygen saturation values based on the second discrete signal values, the oxygen saturation values corresponding with oxygen saturation of the patient's blood;

adjusts the first oxygen saturation values to produce first adjusted oxygen saturation values by smoothing the first oxygen saturation values, wherein the digital signal processor determines to use said smoothing based at least in part on the first amount of noise;

adjusts the second oxygen saturation values to produce second adjusted oxygen saturation values by applying a curve-fit to the second oxygen saturation values, wherein the digital signal processor determines to use the curve-fit based at least in part on the second amount of noise; and

outputs the first and second adjusted oxygen saturation values to a display.

20. The system of claim 19, wherein the digital signal processor computes a first signal confidence based on the first amount of noise and a second signal confidence based on the second amount of noise.

21. The system of claim 20, wherein the digital signal processor determines to use said smoothing based on the first signal confidence, and wherein the digital signal processor determines to use the curve-fit based on the second signal confidence.

22. The system of claim 20, wherein the digital signal processor computes the first signal confidence based at least in part by measuring pulse rate and signal strength of the first discrete signal values.

23. The system of claim 22, wherein the digital signal processor computes a higher value of the first signal confidence when the signal strength is within an expected range for the measured pulse rate or a lower value of the first signal confidence when the signal strength is outside an expected range for the measured pulse rate.

24. The system of claim 22, wherein the digital signal processor performs said smoothing by determining a moving weighted average of the first oxygen saturation values.

25. The system of claim 22, wherein the curve-fit is a non-linear curve fit.

26. The system of claim 25, wherein the non-linear curve fit is an exponential curve fit.

27. A system comprising:

an analog signal conditioning circuit that receives signals corresponding to attenuated light transmitted through body tissue of a patient, the analog signal conditioning circuit comprising an analog-to-digital conversion circuit that provides first discrete signal values at a first time and second discrete signal values at a second time subsequent to the first time;

a digital-to-analog conversion circuit that provides control signals to emitter current drivers to cause the light to transmit through the body tissue; and

a processor that:

receives the first discrete signal values and the second discrete signal values from the analog signal conditioning circuit;

calculates first physiological parameter values based on the first discrete signal values and second physiological parameter values based on the second discrete signal values;

calculate a first signal confidence based on first noise in the first discrete signal values and a second signal confidence based on second noise in the second discrete signal values;

adjusts the first physiological parameter values to produce first adjusted physiological parameter values by smoothing the first physiological parameter values, wherein the processor uses said smoothing based at least in part on the first signal confidence;

adjusts the second physiological parameter values to produce second adjusted physiological parameter values by applying filtering to the second physiological parameter values, wherein the processor uses said filtering based at least in part on the second signal confidence; and

provide the first and second adjusted physiological parameter values for output to a display.

28. The system of claim 27, wherein the filtering comprises non-linear filtering.

29. The system of claim 28, wherein the filtering comprises exponential filtering.

30. The system of claim 27, wherein the smoothing comprises averaging.

31. The system of claim 27, wherein the filtering comprises curve-fitting.

32. The system of claim 27, wherein the processor uses said smoothing instead of said filtering for the first physiological parameter values based on the first signal confidence being lower than the second signal confidence.

33. The system of claim 32, wherein the processor uses said filtering instead of said smoothing for the second physiological parameter values based on the second signal confidence being higher than the first signal confidence.

34. The system of claim 27, wherein the processor calculates third physiological parameter values and adjusts the third physiological parameter values with a combination of smoothing and filtering.

35. The system of claim 34, wherein the combination of smoothing and filtering comprises applying a single equation that incorporates both averaging and curve fitting.

\* \* \* \* \*

|                |  |         |            |
|----------------|--|---------|------------|
| 专利名称(译)        | 可变指示估计器  |         |            |
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| [标]申请(专利权)人(译) | 梅西莫股份有限公司  |         |            |
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| 其他公开文献         | US10357206   |         |            |
| 外部链接           | <a href="#">Espacenet</a> <a href="#">USPTO</a>  |         |            |

#### 摘要(译)

一种可变指示估计器，其确定表示一组输入数据的输出值。例如，估计器可以将输入数据减少到期望信号的估计，选择时间，并根据估计和时间确定输出值。在一个实施例中，使用一个或多个可调信号置信度参数来选择时间，确定沿估计的位置将计算输出值。通过改变参数，输出值的特征是可变的。例如，当输入信号置信度低时，调整参数使得输出值是输入信号的平滑表示。当输入信号置信度高时，调整参数以使输出值对输入信号具有更快和更准确的响应。

