

Description

TECHNICAL FIELD

The invention relates to an improved method and arrangement for processing radio signals that vary over a wide dynamic range, the radio signals carrying information in a signal amplitude and in a corresponding phase angle. According to the present invention there is provided a method and an arrangement as defined in claim 1 and 2 respectively.

BACKGROUND ART

It is always possible to represent an arbitrary radio signal as a sequence of composite (complex) vectors. The real and imaginary parts of the vector sequence correspond to bipolar amplitude modulation (double side band suppressed carrier AM) of a cosine and sine carrier wave respectively (quadrature carriers). When wishing to process a radio signal numerically using digital arithmetic implemented in either specific hardware logic or in software on a computer, microprocessor or some other programmable apparatus, it is first necessary to convert the signal in to numerical form with the aid of a A/D-converter (Analogue to Digital converter).

One common method of achieving this is to first resolve the radio signal into its real and imaginary complex vector part, by correlation with locally generated cosine and sine waves in two balanced mixers, and then to digitalize the two results by means of A/D-conversion. Sometimes there is used a variation of this technique, in which the radio signal is sampled pairwise, separated by one quarter period of its centre frequency. This so-called quadrature sampling technique combines the functions of sampling and A/D-conversion with resolution in real and imaginary parts.

An article: "A True Logarithmic Amplifier for Radar IF Applications" by W.L. Barber and E.R. Brown, published in IEEE Journal of Solide-State Circuits, Vol.SC-15, No.3, June 1980, discloses different arrangements for handling radio signals with large dynamic range. One of the arrangements comprises a number of identical, series coupled limiting amplifiers. The output of the different amplifiers are summated, giving a logarithmic value of the momentary insignal value. The arrangement has the drawback that a compromise must be reached between the accuracy required and the gain of each amplifier. Another arrangement, also using series coupled limiting amplifiers, has a detector connected to each of the amplifiers. The output of these detectors are summated to provide a logarithmic amplitude output, but no phase in-

formation is obtained. A further logarithmic amplifier is disclosed, comprising series coupled so called dual-gain stages. Each stage consists of a limiting amplifier in parallel with a unity gain amplifier, the outputs of which are summated. This type of logarithmic amplifier gives a phase value for the insignal but has the drawback of being complicated. The logarithmic amplifiers will be discussed more in detail below.

DISCLOSURE OF INVENTION

The aforesaid, known solutions have practical limitations with respect to the possibilities of handling the dynamic ranges of the signals. Despite the absence of an input signal, the arrangement used in accordance with the first method, in which balanced mixers are used as correlators, does not necessarily produce a zero (0) volts, output signal. The output signal will typically have a D.C. off-set of some few millivolts or some tens of millivolts. At the same time, the acceptable, maximum signal level of the available supply voltage is limited to, for instance, +2.5 volts or, in the case of diode-ring mixers, perhaps to a still lower level of, for instance, +250 mV. The dynamic range for which the signal is, on one hand, much higher than the D.C offset (mixer imbalance) and, on the other hand, lower than the saturation level, may be as small as 20 dB (decibel). This then requires the introduction of some form of automatic amplification control, in order to maintain the signal level of the mixer in the optimum range. In the case of a receiver which must necessarily accept random transmission of data in the form of bursts from different transmitters, it is not possible, however, to predict the level of amplification required, when applying this method.

A further drawback, applicable to both of the aforesaid methods, resides in limited resolution during the A/D-conversion process. Assume that an A/D-converter is able to represent the whole of the signal level range. Further assume that the highest signal level may be equal to the supply voltage, e.g. 5 volts. An LSB-bit (Least Significant Bit) then corresponds to $5/256$ volts, i.e. approximately 20 millivolt. Consequently, a signal beneath 20 mV will remain totally undiscovered, while a signal of 320 mV will only be digitalized to a resolution of 4 bits, which is perhaps insufficient for subsequent signal processing. If a 4 bit resolution is nevertheless acceptable, the range in which the signals can be processed will be 16:1 or 24dB, which is a very poor dynamic range in the case of radio applications.

Radar receivers are typical examples of systems in which it is impractical to use automatic amplification control for the purpose of maintaining

the receiver output within narrow limits, this impracticability being due to a number of unknown parameters, for instance such parameters as the distance to the reflecting object, the size of said object and the duration of the pulse. Because of this a radar receiver will normally operate with a chain of intermediate frequency amplifiers known as "logarithmic amplifiers". Such an arrangement comprises a plurality of sequentially saturating, cascade-connected amplifiers each being provided with an amplitude detector (rectifier) whose output signals are intended to be added together. The arrangement functions in the following manner: In the case of the weakest input signal levels, it is solely the detector which is located at the end of the chain which will receive a signal whose level of amplification is sufficient for the detector itself to produce an output signal. This ability increases with increasing input signal levels, until the amplifying stage concerned is saturated. At this stage, and with correct selection of amplification for each amplifying stage, the preceding amplifying stage in the chain will begin to receive a signal which is sufficiently strong for detection purposes and therewith takes over the contribution to the output signal. For each X dB increase in input signal level, where X is $20 \log_{10}$ of the voltage amplification in each stage, the saturation point is moved rearwardly one stage in the chain, the net detected output signal therewith increasing by one unit. The net detected output signal is thus followed by an approximately rectilinear relationship with the logarithm on the input signal level. The dynamic range for which this coincides is limited solely by the number of amplifying stages and the thermal noise. The method of digitalizing the detected output signal for subsequent numeric processing of the signal in an arrangement according to the foregoing is insufficient when handling arbitrary radio signals, since the complex vector nature of the arbitrary radio signal will be lost in such a sequential detecting process.

The method and arrangement solving said problems are characterized by the patent claims and involve the introduction of a further digitalizing process which operates on the saturated output of the last amplifying stage in an amplifier chain in accordance with the foregoing, extracting the vector information which otherwise would be lost. This procedure is followed by a multiple of numeric operations on the two digital quantities, in order to restore the complete vector characteristic of the signal. This can be effected with the aid of hardware logic or with programmable digital signal processors (microprocessors). The inventive digitalizing arrangement, intended for processing composite signals having a large dynamic range, thus includes a logarithmic amplifying chain similar to

the kind used in radar receivers and in which the detected output signal from the amplifier is digitalized in a first A/D-converter, whereafter a second A/D-converter digitalizes the angle or phase information of the signal. The phase information is retained by utilizing a carefully configured chain of saturating amplifiers, and is available on the saturated output of the last amplifier stage, at which point the signal has obtained a constant level and all variations in amplitude have therewith been eliminated. The exact method in which phase information is extracted in the form of a numeric quantity is not an objective of the present invention and will therefore not be described in this document.

The advantages afforded by the inventive method and inventive arrangement reside in the solution of a troublesome problem within the field of radio communications, in a technically uncomplicated manner, therewith achieving high precision at low costs.

BRIEF DESCRIPTION OF DRAWING

An arrangement according to the invention will now be described in more detail with reference to an exemplifying embodiment thereof illustrated in the accompanying drawing, in which

Figure 1 is a block schematic of the inventive arrangement, and

Figure 2 is a simplified, schematic illustration of one variant of an amplifier chain according to Figure 1.

BEST MODE FOR CARRYING OUT THE INVENTION

Described in the following is a novel method of digitalizing composite signals and an arrangement for carrying out the method. A complex number can be expressed either in cartesian (x,y) form or in polar (R, THETA) form. Translation between these two forms can be effected readily with the aid of the equations $X = R \cos(\text{THETA})$; $Y = R \sin(\text{THETA})$.

The log polar form (r, THETA) in which $r = \log(R)$ can be advantageously used as an alternative to the two forms aforesaid. The following transformation will then apply: $(x,y) = \exp(r + j \text{THETA})$; $r - j(\text{THETA}) = \log(x,y)$.

These equations show that when having a value on the logarithm of the amplitude of the complex vector (r) and a value on the angle (THETA) of the complex vector, it is possible to restore the cartesian components of the complex vector when so desired.

The inventive digitalizing arrangement for complex signals having wide dynamic ranges utilizes

the principle illustrated in Figure 1. The signal to be processed is converted to an appropriate intermediate frequency and then applied to the input IN of the first amplifier of an amplifier chain A. Said chain including a number of progressively detecting amplifiers.

Suitable, logarithmic amplifiers are available on the market, in the form of integrated circuits. Each stage of the aforesaid amplifier chain consists of one such circuit of the type SL521 A (Plessey Semiconductors). It is also possible to incorporate all amplifying stages in one single circuit, for instance a Signetics SA 804 circuit.

Connected to each of the output of respective amplifying stages is a detector (a rectifier) which has the form of a diode circuit and which is individual to respective stages. The detector outputs are all connected to a summation circuit S of type LF 157 A (National Semiconductors), in which the values from each detector circuit are summated and produced in the form of a summated signal of the output of the summation circuit. This output is connected to an input of a first rapid analogue/digital converter AD1, for instance a converter of type MP 7683 (Micropower Systems). The LOG-amplitude, quantized to N bits, is produced on the outputs of the A/D-converter and delivered to a first number of inputs of a digital signal processor MP. N must be large enough to cover the desired dynamic range in increments or steps which are sufficiently small for the application concerned. For example, if a signal variation range of 128 dB is to be covered and $N=8$ bits, the size of the quantizing steps will be $128/28=0.5$ dB. Said steps size must be small enough to reduce the quantization noise to a level adapted to the application concerned. Quantization is a known technique which lies outside the concept of the present invention and which will not therefore be described in detail in this document.

The signal produced on an output C of the amplifying chain is so strongly amplified as it appears hardlimited (clipped), i.e. the amplifiers are so amplified that the signal is converted into a two-level signal, a square-wave of alternating high or low level. This signal retains the phase-angle information of the original signal when timing the transitions between the two signal levels. The exact method in which phase angle information is extracted in numeric form constitutes no part of this invention, but can for instance be effected with the aid of an appropriate phase detector which is operative to compare the limited square wave with a reference square wave and then to produce an analogue voltage which is proportional to the phase difference, subsequent to which it is necessary to digitalize the signal in an analogue/digital converter. The signal produced on the output C of the last

amplifier stage is applied to an input of a second A/D-converter AD2, in which the phase information of the signal is quantized to M bits and transmitted from the outputs of the A/D-converter to a second multiple of inputs on the digital signal processor MP. This processor may be of the type TMS 320 C 25 (Texas Instruments) or some corresponding processor. There can be used any microprocessor whatsoever which is capable of effecting the log-polar/cartesian transformation at a speed rapid enough for the application concerned, when this is the form required for further processing. The cartesian signal components are produced on the outputs of the microprocessor, as will be seen from Figure 1. In the case of the arrangement illustrated in Figure 1, it is necessary to limit the bandwidth of the amplifier chain, in order to prevent the generation of excessive noise. Consequently, propagation of the signal through the amplifier chain will be delayed, resulting in a continuous delay in the contribution from each of the detector steps. In order to prevent the introduction of disturbances in the case of rapid variations in signal amplitude, it may be necessary to correct for this relative delay prior to the summation of said values.

Since an important feature of the inventive arrangement resides in the digitalizing of momentary envelope changes in the signal, the aforesaid relative delay may be compensated by including in the system a delay line DL with taps at given time distances. Figure 2 shows such a delay line connected to the detector outputs of the amplifier chain. The taps T1-Tn can be adjusted automatically so as to compensate for the delay occurring in the amplifiers. The tap output signals are then summated and delivered to the first A/D-converter AD1. Examples of other compensation methods include the use of switched capacitors or some other CCD-technique (Charge Coupled Device). Alternatively, the output signals from each amplifier stage or groups of amplifying stages may be digitalized separately with the aid of sampling clock signals and the individual values then added together digitally. Timing of the synchronizing and sampling processes is effected in a known manner with the aid of system clock CL, indicated purely schematically in the drawing.

When applying known techniques, amplitude information is extracted very seldomly, and then only for the purpose of establishing the long-distribution quality of the signal and not with the intention of restore the vector characteristic of the signal, as is the intention with the inventive arrangement. As will be understood from the foregoing, in order to achieve this it is necessary to digitalize the signal amplitude and phase angle synchronously at the same sampling rate, and to keep the values together in pairs for each sample,

with the intention of restoring completely the instantaneous, composite vector sequency of the radio signal for use in the continued processing of the signal.

Claims

1. A method for processing radio signals that vary over a wide dynamic range, the radio signals carrying information in a signal amplitude and in a corresponding phase angle, said method comprising the steps of:

- amplifying the radio signals in a plurality of serial amplifying stages (A),
- detecting (D) the output amplitude of each amplifying stage in each a rectifying detector,
- summing (S) the detected (D) outputs of each amplifying stage together, in order to provide signals proportional to the logarithm of the amplitude components of the radio signals and
- analogue-to-digital converting (AD1) said signals proportional to the logarithm of the amplitude components of the radio signals, in order to provide digital converted logarithmic amplitude signals,

characterized in that the method comprises the further steps of:

- generating analogue-to-digital converted signals corresponding to the phase angle of signals from the saturated output (C) of the last of said serial amplifying stages, in order to provide digital phase signals and
- processing (MP) pairwise said digital converted logarithmic amplitude signals and said digital phase angle signals in order to provide restored digital signals (x,y) representative of the radio signals.

2. An arrangement for processing radio signals that vary over a wide dynamic range, the radio signals carrying information in a signal amplitude and in a corresponding phase angle, said arrangement comprising:

- a logarithmic amplifier (A) having a plurality of series coupled amplifiers, the logarithmic amplifier receiving the radio signals on its input (IN) and amplifying the radio signals in a plurality of serial stages,
- a rectifying detector circuit (D) connected to the output of each of the series coupled amplifiers, detecting the output amplitude of the respective amplifier,
- a summation circuit (S) connected to the output of each detector circuit (D), sum-

ming the outputs of said detector circuits (D) in order to provide signals proportional to the logarithm of the amplitude of the radio signals and

- a first analogue-to-digital converter (AD1) converting said signals proportional to the logarithm of the amplitude in order to provide digital converted logarithmic amplitude signals,

characterized in that the arrangement further comprises:

- a second analogue-to-digital converter (AD2) connected to the saturated output (C) of the last of said series coupled amplifiers, the second analogue-to-digital converter (AD2) generating digital phase angle signals corresponding to the phase angle of the signals on said saturated output (C) and
- a digital processor (MP) connected to the output of both the first and the second analogue-to-digital converter (AD1,AD2), said digital processor (MP) pairwise processing the corresponding digital signals from the first and the second analogue-to-digital converter (AD1,AD2), in order to provide restored digital signals (x,y) representative of the radio signals.

3. An arrangement for processing radio signals that vary over a wide dynamic range according to claim 2,

characterized in that the outputs of the detector circuits (D) are connected to a delay line (DL) having output taps (T1-Tn) which are connected to the summation circuit (S), the delay line compensating for a signal delay caused by the series coupled amplifiers and the detector circuits (D) in the logarithmic amplifier (A).

Patentansprüche

1. Verfahren zur Verarbeitung von Funksignalen, die über einen weiten dynamischen Bereich variieren, wobei die Funksignale in einer Signalamplitude und in einem entsprechenden Phasenwinkel Informationen tragen, wobei das Verfahren die folgenden Schritte umfaßt:

- Verstärken der Funksignale in einer Vielzahl von in Reihe angeordneten verstärkenden Stufen (A);
- Erfassen (D) der Ausgangsamplitude jeder verstärkenden Stufe in einem jeweiligen gleichrichtenden Detektor;
- Zusammenaddieren (S) der erfaßten (D) Ausgänge von jeder verstärkenden Stufe, um Signale bereitzustellen, die proportional zu dem Logarithmus der Amplituden-



with the intention of restoring completely the instantaneous, composite vector sequency of the radio signal for use in the continued processing of the signal.

Claims

1. A method for processing radio signals that vary over a wide dynamic range, the radio signals carrying information in a signal amplitude and in a corresponding phase angle, said method comprising the steps of:

- amplifying the radio signals in a plurality of serial amplifying stages (A),
- detecting (D) the output amplitude of each amplifying stage in each a rectifying detector,
- summing (S) the detected (D) outputs of each amplifying stage together, in order to provide signals proportional to the logarithm of the amplitude components of the radio signals and
- analogue-to-digital converting (AD1) said signals proportional to the logarithm of the amplitude components of the radio signals, in order to provide digital converted logarithmic amplitude signals,

characterized in that the method comprises the further steps of:

- generating analogue-to-digital converted signals corresponding to the phase angle of signals from the saturated output (C) of the last of said serial amplifying stages, in order to provide digital phase signals and
- processing (MP) pairwise said digital converted logarithmic amplitude signals and said digital phase angle signals in order to provide restored digital signals (x,y) representative of the radio signals.

2. An arrangement for processing radio signals that vary over a wide dynamic range, the radio signals carrying information in a signal amplitude and in a corresponding phase angle, said arrangement comprising:

- a logarithmic amplifier (A) having a plurality of series coupled amplifiers, the logarithmic amplifier receiving the radio signals on its input (IN) and amplifying the radio signals in a plurality of serial stages,
- a rectifying detector circuit (D) connected to the output of each of the series coupled amplifiers, detecting the output amplitude of the respective amplifier,
- a summation circuit (S) connected to the output of each detector circuit (D), sum-

ming the outputs of said detector circuits (D) in order to provide signals proportional to the logarithm of the amplitude of the radio signals and

- a first analogue-to-digital converter (AD1) converting said signals proportional to the logarithm of the amplitude in order to provide digital converted logarithmic amplitude signals,

characterized in that the arrangement further comprises:

- a second analogue-to-digital converter (AD2) connected to the saturated output (C) of the last of said series coupled amplifiers, the second analogue-to-digital converter (AD2) generating digital phase angle signals corresponding to the phase angle of the signals on said saturated output (C) and
- a digital processor (MP) connected to the output of both the first and the second analogue-to-digital converter (AD1,AD2), said digital processor (MP) pairwise processing the corresponding digital signals from the first and the second analogue-to-digital converter (AD1,AD2), in order to provide restored digital signals (x,y) representative of the radio signals.

3. An arrangement for processing radio signals that vary over a wide dynamic range according to claim 2,

characterized in that the outputs of the detector circuits (D) are connected to a delay line (DL) having output taps (T1-Tn) which are connected to the summation circuit (S), the delay line compensating for a signal delay caused by the series coupled amplifiers and the detector circuits (D) in the logarithmic amplifier (A).

Patentansprüche

1. Verfahren zur Verarbeitung von Funksignalen, die über einen weiten dynamischen Bereich variieren, wobei die Funksignale in einer Signalamplitude und in einem entsprechenden Phasenwinkel Informationen tragen, wobei das Verfahren die folgenden Schritte umfaßt:

- Verstärken der Funksignale in einer Vielzahl von in Reihe angeordneten verstärkenden Stufen (A);
- Erfassen (D) der Ausgangsamplitude jeder verstärkenden Stufe in einem jeweiligen gleichrichtenden Detektor;
- Zusammenaddieren (S) der erfaßten (D) Ausgänge von jeder verstärkenden Stufe, um Signale bereitzustellen, die proportional zu dem Logarithmus der Amplituden-



- Komponenten der Funksignale sind; und
- Durchführen einer Analog-Digital-Wandlung (AD1) der Signale, die proportional zum Logarithmus der Amplitudenkomponenten der Funksignale sind, um in Digitalform gewandelte logarithmische Amplitudensignale bereitzustellen;

dadurch gekennzeichnet, daß das Verfahren außerdem die weiteren Schritte umfaßt:

- Erzeugen von analog-digital-gewandelten Signalen entsprechend dem Phasenwinkel von Signalen von dem gesättigten Ausgang (C) der letzten der in Reihe angeordneten verstärkenden Stufen, um digitale Phasensignale bereitzustellen; und
- paarweise Verarbeitung (MP) der in Digitalform gewandelten logarithmischen Amplitudensignale und der digitalen Phasenwinkelsignale, um wiederhergestellte digitale Signale (x, y) bereitzustellen, die die Funksignale darstellen.

2. Anordnung zur Verarbeitung von Funksignalen, die in einen weiten dynamischen Bereich variieren, wobei die Funksignale in einer Signalamplitude und in einem entsprechenden Phasenwinkel Informationen tragen, wobei die Anordnung umfaßt:

- einen logarithmischen Verstärker (A) mit einer Vielzahl von in Reihe geschalteten Verstärkern, wobei der logarithmische Verstärker die Funksignale an seinem Eingang (EIN) empfängt und die Funksignale in einer Vielzahl von in Reihe angeordneten Stufen verstärkt;
- eine gleichrichtende Detektorschaltung (D), die mit dem Ausgang von jedem der in Reihe geschalteten Verstärker verbunden ist und die Ausgangsamplitude des jeweiligen Verstärkers erfaßt;
- eine Summationsschaltung (S), die mit dem Ausgang jeder Detektorschaltung (D) verbunden ist und die Ausgänge der Detektorschaltungen (D) summiert, um Signale bereitzustellen, die zu dem Logarithmus der Amplitude der Funksignale proportional sind; und
- einen ersten Analog-Digital-Wandler (AD1), der die Signale, die proportional zu dem Logarithmus der Amplitude sind, wandelt, um in Digitalform gewandelte logarithmische Amplitudensignale bereitzustellen;

dadurch gekennzeichnet, daß die Anordnung außerdem umfaßt:

- einen zweiten Analog-Digital-Wandler (AD2), der mit dem gesättigten Ausgang

(C) des letzten Verstärkers der in Reihe geschalteten Verstärker verbunden ist, wobei der zweite Analog-Digital-Wandler (AD2) digitale Phasenwinkelsignale entsprechend dem Phasenwinkel der Signale auf dem gesättigten Ausgang (C) erzeugt; und

- einen digitalen Prozessor (MP), der mit dem Ausgang sowohl des ersten als auch des zweiten Analog-Digital-Wandlers (AD1, AD2) verbunden ist, wobei der digitale Prozessor (MP) die entsprechenden digitalen Signale von dem ersten und dem zweiten Analog-Digital-Wandler (AD1, AD2) paarweise verarbeitet, um wiederhergestellte digitale Signale (x, y) bereitzustellen, die die Funksignale darstellen.

3. Anordnung zur Verarbeitung von Funksignalen, die in einem weiten dynamischen Bereich variieren, gemäß Anspruch 2, dadurch gekennzeichnet, daß die Ausgänge der Detektorschaltungen (D) mit einer Verzögerungsleitung (DL) verbunden sind, die Ausgangsabgriffe (T1-Tn) aufweist, die mit der Summationsschaltung (S) verbunden sind, wobei die Verzögerungsleitung eine Signalverzögerung kompensiert, die von den in Reihe geschalteten Verstärkern und von den Detektorschaltungen (D) in dem logarithmischen Verstärker (A) herrührt.

Revendications

1. Un procédé de traitement de signaux radioélectriques qui varient sur une dynamique étendue, ces signaux radioélectriques acheminant de l'information qui est contenue dans une amplitude de signal et dans un angle de phase correspondant, ce procédé comprenant les étapes suivantes :
- on amplifie les signaux radioélectriques dans un ensemble d'étages amplificateurs série (A),
 - on détecte (D) l'amplitude de sortie de chaque étage amplificateur dans un détecteur redresseur respectif,
 - on somme ensemble (S) les signaux de sortie détectés (D) de chaque étage amplificateur, dans le but de produire des signaux proportionnels au logarithme des composantes d'amplitude des signaux radioélectriques, et
 - on effectue une conversion analogo-numérique (AD1) des signaux proportionnels au logarithme des composantes d'amplitude des signaux radioélectriques,



dans le but de produire des signaux d'amplitude logarithmique convertis sous forme numérique,

caractérisé en ce que le procédé comprend les étapes supplémentaires suivantes :

- on génère des signaux convertis de la forme analogique à la forme numérique, correspondant à l'angle de phase de signaux provenant de la sortie saturée (C) du dernier des étages amplificateurs série, dans le but de produire des signaux de phase numériques,
- on traite (MP) par paires les signaux d'amplitude logarithmique convertis sous forme numérique et les signaux d'angle de phase numériques, dans le but de produire des signaux numériques rétablis (x, y) représentatifs des signaux radioélectriques.

2. Un dispositif pour traiter des signaux radioélectriques qui varient sur une dynamique étendue, ces signaux radioélectriques acheminant de l'information qui est contenue dans une amplitude de signal et dans un angle de phase correspondant, ce dispositif comprenant :

- un amplificateur logarithmique (A) comprenant un ensemble d'amplificateurs connectés en série, l'amplificateur logarithmique recevant les signaux radioélectriques sur son entrée (IN) et amplifiant les signaux radioélectriques dans un ensemble d'étages série,
- un circuit détecteur redresseur (D) connecté à la sortie de chacun des amplificateurs connectés en série, qui détecte l'amplitude de sortie de l'amplificateur respectif,
- un circuit de sommation (S), connecté à la sortie de chaque circuit détecteur (D), qui fait la somme des signaux de sortie des circuits détecteurs (D), dans le but de produire des signaux proportionnels au logarithme de l'amplitude des signaux radioélectriques, et
- un premier convertisseur analogique-numérique (AD1) qui convertit les signaux proportionnels au logarithme de l'amplitude, dans le but de produire des signaux d'amplitude logarithmique numériques convertis,

caractérisé en ce que le dispositif comprend en outre :

- un second convertisseur analogique-numérique (AD2) connecté à la sortie saturée (C) du dernier des amplificateurs connectés en série, le second convertisseur analogique-numérique (AD2) géné-

rant des signaux d'angle de phase numériques qui correspondent à l'angle de phase des signaux présents sur la sortie saturée (C), et

- un processeur numérique (MP) qui est connecté à la sortie des premier et second convertisseurs analogique-numérique (AD1, AD2), ce processeur numérique (MP) traitant par paires les signaux numériques correspondants qui proviennent des premier et second convertisseurs analogique-numérique (AD1, AD2), dans le but de produire des signaux numériques rétablis (x, y) qui sont représentatifs des signaux radioélectriques.

3. Un dispositif pour traiter des signaux radioélectriques qui varient sur une dynamique étendue selon la revendication 2,

caractérisé en ce que les sorties des circuits détecteurs (D) sont connectées à une ligne à retard (DL) ayant des prises de sortie (T1-Tn) qui sont connectées au circuit de sommation (S), la ligne à retard compensant un retard de signal qui est occasionné par les amplificateurs connectés en série et les circuits détecteurs (D) dans l'amplificateur logarithmique (A).