

(19)



Europäisches Patentamt

European Patent Office

Office européen des brevets



(11)

EP 0 464 217 B1

(12)

EUROPEAN PATENT SPECIFICATION

(45) Date of publication and mention
of the grant of the patent:

12.06.1996 Bulletin 1996/24

(21) Application number: **91902738.3**

(22) Date of filing: **18.01.1991**

(51) Int Cl.⁶: **H04S 1/00, H04R 5/033**

(86) International application number:
PCT/JP91/00057

(87) International publication number:
WO 91/11080 (25.07.1991 Gazette 1991/17)

(54) **APPARATUS FOR REPRODUCING ACOUSTIC SIGNALS**

GERÄT ZUR WIEDERGABE VON TONSIGNALEN

APPAREIL DE REPRODUCTION DE SIGNAUX ACOUSTIQUES

(84) Designated Contracting States:
DE FR GB NL

(30) Priority: **19.01.1990 JP 8514/90**
19.01.1990 JP 8520/90

(43) Date of publication of application:
08.01.1992 Bulletin 1992/02

(60) Divisional application: **95104929.5**

(73) Proprietor: **SONY CORPORATION**
Tokyo 141 (JP)

(72) Inventors:
• **INANAGA, Kiyofumi**
Shinagawa-ku Tokyo 141 (JP)

- **SOGAWA, Hiroyuki**
Shinagawa-ku Tokyo 141 (JP)
- **IIDA, Yasuhiro**
Shinagawa-ku Tokyo 141 (JP)
- **YABE, Susumu**
Shinagawa-ku Tokyo 141 (JP)

(74) Representative: **Ayers, Martyn Lewis Stanley et al**
J.A. KEMP & CO.
14 South Square
Gray's Inn
London WC1R 5LX (GB)

(56) References cited:
WO-A-89/03632 **JP-A- 1 121 000**
JP-A- 5 165 901 **JP-A-54 109 401**
JP-A-58 116 900

EP 0 464 217 B1

Note: Within nine months from the publication of the mention of the grant of the European patent, any person may give notice to the European Patent Office of opposition to the European patent granted. Notice of opposition shall be filed in a written reasoned statement. It shall not be deemed to have been filed until the opposition fee has been paid. (Art. 99(1) European Patent Convention).

Description

FIELD OF THE INVENTION

The present invention relates to an audio signal binaural reproducing apparatus for reproducing audio signals by means of headphones.

BACKGROUND OF THE INVENTION

A binaural reproducing method has heretofore been known as an approach for providing better direction sensation of sound image or outside head localization sensation when audio signals are reproduced by headphones fitted to the head of a listener so that a pair of headphones are located in the vicinity of both ears.

An audio reproducing system adopting this binaural system preliminarily applies a given signal processing to the audio signals reproduced by headphones as is described in, for example, specification of Japanese Patent Publication Sho 53-283.

The direction sensation of sound image and outside head localization sensation and the like depend upon the differences in volumes, times and phases of sounds listened by left and right ears.

The signal processing aims at causing in an audio output reproduced by the headphones, audio effects equivalent to those caused by the difference in distances between sound sources, that is, speaker systems and right and left ears of a listener and reflections and diffractions in the vicinity of the head of the listener when audio reproducing is performed, for example, by speaker systems remote from the listener. Such a signal processing is performed by convolution-integrating left and right ear audio signals with impulse responses corresponding to the above-mentioned audio effects.

Since the absolute position of the sound image is not changed even if the listener moves or turns his or her head when audio reproducing is performed by speaker systems remote from the listener, the relative direction and position of the sound image that the listener senses are changed. In contrast to this, since the headphones is turned together with the listener's head if the listener turns his or her head when audio reproducing is performed by a binaural method using headphones, the relative direction and position of the sound image which the listener senses are not changed.

If binaural reproducing is performed by using headphones in such a manner, a sound image is created in the head of a listener due to differences in displacement of the sound image relative to a change in direction of the listener's head. Therefore, it is difficult to locate the sound image in front of the listener. Furthermore, the front sound image has a tendency to lift up.

Accordingly, an audio signal reproducing system which detects a change in the direction of the listener's head and changes the modes of the signal processing based upon a result of the detection for providing a good

front localization sensation in headphones has heretofore been proposed as is disclosed in Japanese Unexamined Patent Publication No. Sho 42-227 and Japanese Examined Patent Publication No. 54-19242. In such an audio signal reproducing system, a direction detecting device such as gyrocompass and magnetic needle is provided on the head of the listener. A level adjusting circuit and a delay circuit and the like for processing the audio signals are controlled based upon a result of detection from the direction detecting device so that a sound image sensation which is similar to that of the audio reproducing using speaker systems remote from the listener is obtained.

In the prior art binaural reproducing system in which headphones are provided with a direction detecting device comprising a gyrocompass, an excellent sound image can be obtained by controlling the content of the signal processing which is applied to the audio signals depending upon changes in direction of the listener's head. JP-A-1121 000 discloses on audio signal reproducing apparatus according to the preamble of Claim 1. A similar arrangement is disclosed in JP-A-58 116 900.

In order to control the content of the signal processing applied to the audio signals depending upon a change in direction of the listener's head, it is necessary to preliminarily measure the impulse responses, that is, transfer characteristics corresponding to audio effects given to audio signals of left and right ears for each predetermined rotational angle and to store a great amount of information on the transfer characteristics. The information is read from the storing means depending upon the change in direction of the head. The audio signal will be subjected to a necessary convolution-integration processing in real-time.

The present invention was made with a view to overcome such drawbacks.

It is an object of the present invention to provide an audio signal reproducing apparatus having a simple structure using storing means having a low storing capacity which is capable of performing a binaural reproduction for providing a very natural localization of a sound image in which the positions of virtual sound sources are not changed by headphones even if a listener moves by reducing the amount of information on transfer characteristics from virtual sound sources necessary for binaural reproduction of audio signals with the headphones to both ears of the listener.

An audio signal reproducing apparatus of the present invention comprises means for detecting the rotational angular position of the head of a listener corresponding to the movement of the head of the listener relative to virtual sound sources;

means for storing transfer characteristics information representative of transfer characteristics of direct sounds from virtual sound sources to each ear of the listener for predetermined rotational angular positions; and

audio signal processing means for forming, based upon the transfer characteristics information stored in said storing means, transfer characteristic information appropriate to the rotational angular position of the head and for processing left and right channel audio signals using said transfer characteristic information, whereby audio signals which have been processed by said audio signal processing means are reproduced as sounds by headphone means; characterised in that:

said means for storing stores transfer characteristics information of predetermined angular positions in a quadrant and said audio signal processing means derives information for the other quadrants from the stored information.

The present invention will be further described hereinafter with reference to the following description of exemplary embodiments and the accompanying figures, in which:

Fig. 1 is a block diagram schematically showing the structure of an audio signal reproducing apparatus of the present invention;

Fig. 2 is a time chart schematically showing signals supplied to a operation unit of the audio signal reproducing apparatus;

Fig. 3 is a schematic diagram illustrating the distance and the angle calculated by the operation unit of the audio signal reproducing apparatus;

Fig. 4 is a view for explaining the information on the transfer characteristics stored in a storing circuit of the operation unit in the audio signal reproducing apparatus;

Fig. 5 is a plan view showing the relative positional relation between virtual sound sources and a listener for explaining the operation of binaural reproducing performed by the audio signal reproducing apparatus; and

Fig. 6 is a block diagram schematically showing the other structure of the audio signal reproducing apparatus of the present invention.

BEST MODE FOR EMBODYING THE INVENTION

The audio signal reproducing apparatus of the present invention comprises a headphone set 10 which is fitted over the head M of a listener P and a pair of headphones 2L and 2R are supported by a head band 1 so that they are located in the vicinity of left and right ears of the listener P, respectively as shown in Fig. 1.

Two sliders 4L and 4R from which support arms 3L and 3R, respectively project are slidably mounted on the head band 1 of the headphone set 10. A pair of signal detectors 5L and 5R which detect a position detection reference signal emitted from a reference signal source 11 are provided at the tip ends of the support arms 3L and 3R, respectively. That is, the pair of signal detectors

5L and 5R are provided on the tip ends of the support arms 3L and 3R projectedly formed on the sliders 4L and 4R which are slidably mounted on the head band 1 so that they are supported in positions remote from the head band 1 and the pair of headphones 2L and 2R, that is the main body of the headphone set.

In the present embodiment, the reference signal source 11 comprises an ultrasonic signal source 12 and an ultrasonic speaker 13 for generating an ultrasonic signal from the ultrasonic signal source 12 as a reference signal. Each of the pair of signal detectors 5L and 5R which receive the reference signal comprises an ultrasonic microphone.

An ultrasonic wave generated from the ultrasonic speaker 13, that is, the position detection reference signal is a burst wave in which an ultrasonic wave having a given level is intermittently generated for a given period of time as shown at A in Fig. 2, or an ultrasonic wave, the phase of which may be detected like a so-called level modulated wave, the level of which changes in a given circle.

The pair of signal detectors 5L and 5R provided on the headphone set 10 detects the ultrasonic position detection reference signal generated from the ultrasonic speaker 13 and generate respective detection signals shown at B and C in Fig. 2, each having a time lag depending upon the relative positional relation between the listener P and the ultrasonic speaker 13.

Since the pair of signal detectors 5L and 5R are supported by the support arm 3L and 3R in positions remote from the main body of the headphone set 10 while they are mounted on the tip ends of the support arms 3L and 3R in positions remote from the main body of the headphone set 10 while they are mounted on the tip ends of the support arms 3L and 3R which project from the sliders 4L and 4R, respectively slidably mounted on the head band 1 and, the head band 1 and the pair of headphones 2L and 2R, that is, the main body of the headphone set is fitted on the head M of the listener P, they can detect the ultrasonic wave generated from the ultrasonic speaker 13, that is, the position detection reference signal stably and accurately without being located behind the head P of the listener P even if the listener P moves or rotates his head P. The pair of the signal detectors 5L and 5R can be adjusted to a position optimal for detecting the detection reference signal by sliding the sliders 4L and 4R along the head band 1. For example, the optimal positions of the headphones 2L and 2R which are fitted on the head M of the listener P by the head band 1 so that they correspond to the vicinity of the left and right ears depend on the shape and size of the head M of the listener P, that is, have the differences among individuals. Accordingly, the positions of the pair of signal detectors 5L and 5R can be adjusted so that they correspond to the headphones 2L and 2R, respectively.

Each detection signal obtained by these signal detectors 5L and 5R is applied to an operation unit 14.

The operation unit 14 comprises first and second edge detecting circuits 15 and 16, to which the detection signal from the signal detectors 5L and 5R for detecting the position detection reference signal are supplied, respectively and a third edge detecting circuit 17 to which an ultrasonic signal from the ultrasonic signal sources 12, that is, the position detection reference signal is applied.

The first and second edge detecting circuits 15 and 16 detect rise-up edges of the detection signals generated from the signal detectors 5L and 5R, respectively and output pulse signals shown at D and E of Fig. 2 corresponding to the rise-up edges pulse signals generated by the first and second edge detecting circuits 15 and 16 are supplied to a distance calculating circuit 18 and a circuit 19 for detecting the time difference between both ears. The third edge detecting circuit 17 detects the rise-up edge of the ultrasonic signal from the ultrasonic signal source 12 and outputs a pulse signal corresponding to the rise-up edge as shown at F in Fig. 2. A pulse signal obtained by the third edge detection circuit 17 is supplied to the distance calculating circuit 18.

The distance calculating circuit 18 detects the time difference t_1 between pulse signals obtained by the third and first edge detecting circuits 17 and 15 which is represented as ΔT_1 in Fig. 2 and the time difference t_2 between pulse signals obtained by the third and second edge detecting circuits 17 and 16 which is represented as ΔT_2 in Fig. 2 and then calculates the distance ℓ_0 between the ultrasonic speaker 13 and the center of the head M of the listener P represented as ℓ_0 in Fig. 3 based upon the time differences t_1 , t_2 and the sound velocity V.

The sound velocity V may be preliminarily preset as a constant in the distance calculating circuit 18 or alternatively may be changed with changes in atmospheric temperature, humidity and atmospheric pressure and the like. On calculating the distance ℓ_0 , compensation may be conducted for the positional relation between the signal detector 5L and 5R and the center of the head M, the shape and size of the head M.

Signals representative of the distance ℓ_0 , time differences t_1 and t_2 are fed to an angle calculating circuit 20.

The circuit 19 for detecting the time difference between both ears detects the time difference t_3 between the pulse signals generated by the first and second edge detecting circuits 15 and 16, represented as Δ_3 in Fig. 2. A signal representative of the time difference t_3 is fed to the angle calculating circuit 20.

The angle calculating circuit 20 calculates an angle representative of the direction of the head M represented by an arrow θ_0 in Fig. 3 by using the time differences t_1 , t_2 , t_3 , the distance ℓ_0 , the sound velocity V and the radius r of the head M. The angle θ_0 can be determined, for example, by the equation 1 as follows:

$$\theta_0 = \sin^{-1} \{V^2(t_1 + t_2)t_3/4r\ell\} \quad (1)$$

Then, the rotation angle θ of the head M relative to a desired position of a virtual sound source is calculated from information on the angle θ_0 and the distance ℓ_0 representative of the relative positional relationship between a reference position and the listener P by assuming that the position of the ultrasonic speaker 13 be the reference position of the virtual sound source.

Information on the rotation angle of the head of the listener obtained by the angle calculating circuit 20 is provided to a control circuit 21.

In the audio signal reproducing apparatus of this embodiment, the operation unit 14 includes a storing circuit 22 which stores information on transfer characteristics from the virtual sound source to both ears of the listener in at least the first quadrant of the rotational angular position of the head of the listener, for example, information on the transfer characteristics for each angle θ_{11} to θ_{1n} in the first quadrant.

Based upon the current angle position calculated by the angle calculating circuit 20, the control circuit 21 reads the information on the transfer characteristics corresponding to the current angles θ_{11} to θ_{1n} positions from the storing circuit 22 if the current angle position is in the first quadrant in Fig. 4 and reads the transfer characteristics information in which the current angles θ_{21} to θ_{2n} corresponds to the angles θ_{11} to θ_{1n} in the first quadrant from the storing circuit 22 if the current angle position is in the second quadrant in Fig. 4 and read the transfer characteristics information in which the current angles θ_{31} to θ_{3n} corresponds to the angles θ_{11} to θ_{1n} in the first quadrant from the storing circuit 22 if the current angle position is in the third quadrant in Fig. 4 and read the transfer characteristics information in which the current angles θ_{41} to θ_{4n} correspond to the angles θ_{11} to θ_{1n} in the first quadrant from the storing circuit 22 if the current angle position is in the fourth quadrant in Fig. 4 and supplies the read transfer characteristics information to an audio signal processing circuit 23 together with a signal representative of the quadrant in which the current angular position is located.

Since the head of the listener is substantially spherical and rotary symmetric, the transfer characteristics from the virtual sound sources to both ears of the listener can be treated as symmetrical in each quadrant.

Alternatively, in the control circuit 21, two transfer characteristics in the vicinity of the rotational angular position of the head represented by the angular position information may be read from the storing circuit 22 and the information on the transfer characteristics in the current head rotational angular position may be operated by, for example, linear interpolation processing, as described later with reference to Fig. 6.

Left and right channel audio signals S_L and S_R which are outputted from the audio signal source 22 are supplied to the audio signal processing circuit 23.

The audio signal source 24 is an apparatus for outputting given left and right channel audio signals S_L and S_R , such as recording disc playback apparatus or radio communication receivers and the like.

The audio signal processing circuit 23 performs a signal processing which provides the left and right channel audio signals S_L and S_R fed from the audio signal source 24 with a given transfer characteristics from the virtual sound source to the both ears of the listener. The audio signal processing circuit 23 comprises first to sixth switches 25L, 25R, 26L, 26R, 27L and, 27R for switching the signal lines and first to fourth signal processing units 28a, 28b, 28c and 28d.

The first to sixth switches 25L, 25R, 26L, 26R, 27L and 27R are controlled for switching in response to a control signal from the control circuit 21 representative of the quadrant to which the current angular position belongs.

The first and second switches 25L and 25R perform switching of inputs of left and right channel audio signals S_L and S_R fed from the audio signal source 24 and supply the right channel audio signal S_R to the first and second signal processing units 28a and 28b and supply the left channel audio signal S_L to the third and fourth signal processing units 28c and 28d when the current angular position is in the first or third quadrant and supply the left channel audio signal S_L to the first and second signal processing unit 28a and 28b and supply the right channel audio signal S_R to the third and fourth signal processing units 28c and 28d when the current angular position is in the second or fourth quadrant.

The third and fourth switches 26L and 26R perform switching of the output of the left and right channel audio signals E_L and E_R outputted from the audio signal processing circuit 23 and select as a right channel audio signal E_R the output signal of the first adder 29R for adding the output signals of the first and third signal processing unit 28a and 28c and select as a left channel audio signal E_L the output signal of the second adder 29L for adding the output signals of the second and fourth signal processing units 28b and 28d when the current angular position is in the first or third quadrant and select as a right channel audio signal E_R the output signal of the first adder 29L and select as a left channel audio signal E_L the output signal of the second adder 29L when the current angular position is in the second or the fourth quadrant.

The third and fourth switches 26L and 26R perform switching of filters for the left and right channel audio signals E_L and E_R outputted from the audio signal processing circuit 23 and output the left and right audio signals E_L and E_R unswitched when the current angular position is in the second or fourth quadrant and output the audio signals E_L and E_R from which high frequency components have been removed by low pass filters 30L and 30R when the current angular position is in the second or fourth quadrant.

In each of signal processing units 28a, 28b, 28c and

28d, an impulse response representative of the transfer characteristics of the left and right channel audio signals S_L and S_R reproduced from a pair of left and right channel speakers which are virtual sound sources facing to a listener to each ear of the listener is preset based upon information on transfer characteristics supplied from the control circuit 21.

In other words, the first signal processing unit 28a presets the impulse response $\{h_{RR}(t, \theta)\}$ representative of transfer characteristics of the sound reproduced from the right channel audio signal S_R to the right ear. The second signal processing unit 28b presets the impulse response $\{h_{RL}(t, \theta)\}$ representative of the transfer characteristics of the sound reproduced from the right channel audio signal S_R to the left ear. The third signal processing unit 28c presets the impulse response $\{h_{LR}(t, \theta)\}$ representative of transfer characteristics of the sound reproduced from the left channel audio signal S_L to the right ear. The fourth signal processing unit 28d presents the impulse response $\{h_{LL}(t, \theta)\}$ representative of the transfer characteristics of the sound reproduced from the left channel audio signal S_L to the left ear.

When the current angular position of the head of the listener is in the first quadrant, the right channel audio signal S_R is fed the first and second signal processing units 28a and 28b. In the first signal processing unit 28a, the right channel audio signal S_R is subjected to a signal processing of convolution-integration of the impulse response $\{h_{RR}(t, \theta)\}$. In the second signal processing unit 28b, the right channel audio signal S_R is subjected to a signal processing of convolution-integration of the impulse response $\{h_{RL}(t, \theta)\}$.

The left channel audio signal S_L is fed to the third and fourth signal processing units 28c and 28d. In the third signal processing unit 28c, the left channel audio signal S_L is subjected to a signal processing of convolution-integration of the impulse response $\{h_{LR}(t, \theta)\}$. In the second signal processing unit 28d, the left channel audio signal S_L is subjected to a signal processing of convolution-integration of the impulse response $\{h_{LL}(t, \theta)\}$.

The output signals from the first and third signal processing units 28a and 28c are applied to the right channel adder 29R and are added with each other therein. The output signal of the right channel adder 28R is fed as the right channel studio signal E_R via the right channel amplifier 31R to the right channel headphone 2R and reproduced as a sound. The output signals from the second and fourth signal processing units 28b and 28d are applied to the left channel adder 29L and are added with each other therein. The output signal of the left channel adder 29 is fed as the left channel audio signal E_L via the left channel amplifier 31L to the left channel headphone 2L and reproduced as a sound.

When the current angular position of the head of the listener is in the second quadrant, the left and right channels of inputs and outputs are replaced with each other and a processing which is similar to that of the foregoing

first quadrant is performed. Accordingly, a front localization sensation is provided. When the current angular position of the head of the listener is in the third and fourth quadrants a processing which is similar to those of the first and second quadrants is performed. Audio signals E_L and E_R from which high frequency components have been removed from the low pass filters 30L and 30R are outputted. Accordingly, rear localization sensation can be provided.

In the audio signal reproducing apparatus of the thus formed embodiment, information on the transfer characteristics in the rotational angular positions corresponding to the movement of the head of the listener calculated by the angle calculating circuit 20 is formed based upon the information upon the transfer characteristics of the first quadrant stored in the storing circuit 22. By performing a signal processing of the left and right channel audio signals S_L and S_R which responds to changes in transfer characteristics in association with the movement of the listener P and the rotation of the head M in real time in the audio signal processing circuit 23 based upon the transfer characteristics data, good outside head localization sensation and front localization sensation are obtained in which the virtual sound sources are not moved as similarly to the case in which an audio signal is reproduced by a pair of speaker systems S_L and S_R which faces to the listener P and are remote therefrom and with each other as is shown in Figs. 5A, 5B and 5C in which relative positional relations between the virtual sound sources and the listener P are illustrated.

Fig. 5B shows that the listener P has approached to a pair of speaker systems S_L and S_R , that is, virtual sound sources from a position of Fig. 5A. Fig. 4C shows that the listener P rotates his head M toward the right speaker device S_R . By performing a signal processing which can respond in real time to changes in transfer characteristics in association with the movement of the listener and the rotation of the head M as mentioned above in the audio signal reproduction apparatus of the present embodiment, good head outside and front localization sensation in which no virtual sound source is moved can be obtained so that a binaural reproduction which can respond to any conditions of Figs. 5A, 5B and 5C can be performed.

Since it suffices for the audio signal reproducing apparatus of the present invention to store in storing means transfer characteristics information representative of the transfer characteristics from virtual sound sources to a listener of the first quadrant of the rotational angular position of the head of the listener, the amount of the information of the transfer characteristics to be stored in the storing means is small and the storing means having a low storing capacity can be used. The audio signal processing means forms the transfer characteristics information in the rotational angular position represented by a detection output from detecting means for detecting the rotational angular position depending

upon the movement of the head of the listener in accordance with the transfer characteristics information of at least the first quadrant stored in the storing means and processes the left and right channel audio signals for supplying the processed audio signals to the headphone set. Accordingly, a proper binaural reproduction can be performed for providing a very natural sound image localization sensation in which the positions of the virtual sound sources are not moved even if the listener moves.

A second embodiment of an audio signal reproducing apparatus of the present invention will now be described in detail with reference to the drawings.

The audio signal reproducing apparatus of the present invention shown in Fig. 6 comprises a headphone set 40 which is fitted over the head M of a listener P and a pair of headphones 42L and 42R are supported by a head band 41 so that they are located in the vicinity of left and right ears of the listener P, as is similar to the embodiment shown in Fig. 1.

Two sliders 44L and 44R from which support arms 43L and 43R, respectively project are slidably mounted on the head 1 of the headphone set 40. A pair of signal detectors 45L and 45R which detect a position detection reference signal emitted from a reference signal source 51 are provided at the tip ends of the support arms 431 and 43R, respectively. That is, the pair of signal detectors 45L and 45R are provided on the tip ends of the support arms 43L and 43R projectedly formed on the sliders 44L and 44R which are slidably mounted on the head band 51 so that they are supported in positions remote from the head band 51 and the pair of headphones 42L and 42R, that is the main body of the headphone set.

Also in this present embodiment, the reference signal source 51 comprises an ultrasonic signal source 52 and an ultrasonic speaker 53 for generating an ultrasonic signal from the ultrasonic signal source 52 as a reference signal. Each of the pair of signal detectors 45L and 45R which receives the reference signal comprises an ultrasonic microphone.

An ultrasonic wave generated from the ultrasonic speaker 53, that is, the position detection reference signal is a burst wave in which an ultrasonic wave having a given level is intermittently generated for a given period of time as is similar to the first embodiment, or an ultrasonic wave, the phase of which may be detected like a so-called level modulated wave, the level of which changes in a given circle.

The pair of signal detectors 45L and 45R provided on the headphone set 40 detects the ultrasonic position detection reference signal generated from the ultrasonic speaker 53 and generate respective detection signals, each having a time lag depending upon the relative positional relation between the listener P and the ultrasonic speaker 53.

Each detection signal obtained by these signal detectors 45L and 45R is applied to an operation unit 54.

The operation unit 54 comprises first and second edge detecting circuit 55 and 56, to which the detection signal from the signal detectors 45L and 45R for detecting the position detection reference signal are supplied, respectively and a third edge detecting circuit 57 to which an ultrasonic signal from the ultrasonic signal source 52, that is, the position detection reference signal is applied.

The first and second edge detecting circuits 55 and 56 detect rise-up edges of the detection signals generated from the signal detectors 45L and 45R, respectively and output pulse signals corresponding to the rise-up edges pulse signals generated by the first and second edge detecting circuits 55 and 56 are supplied to a distance calculating circuit 58 and a circuit 59 for detecting the time difference between both ears. The third edge detecting circuit 57 detects the rise-up edge of the ultrasonic signal from the ultrasonic signal source 52 and outputs a pulse signal corresponding to the rise-up edge. A pulse signal obtained by the third edge detecting circuit 57 is supplied to the distance calculating circuit 58.

The distance calculating circuit 58 detects the time difference t_1 between pulse signals obtained by the third and first edge detecting circuits 57 and 55 and the time difference t_2 between pulse signals obtained by the third and second edge detecting circuits 57 and 56 and then calculates the distance ℓ_0 between the ultrasonic speaker 53 and the center of the head M of the listener based upon the time differences t_1 , t_2 and the sound velocity V.

Signals representative of the distance ℓ_0 , time differences t_1 and t_2 are fed to an angle calculating circuit 60.

The circuit 59 for detecting the time difference between both ears detects the time difference t_3 between the pulse signals generated by the first and second edge detecting circuits 55 and 56. A signal representative of the time difference t_3 is fed to the angle calculating circuit 60.

The angle calculating circuit 60 calculates an angle θ_0 representative of the direction of the head M by using the time differences t_1 , t_2 , t_3 , the distance ℓ_0 , the sound velocity V and the radius r of the head M similarly to the angle calculating circuit 20 in the first embodiment.

Information on the rotation angular position head of the listener obtained by the angle calculating circuit 60 is provided to an interpolation operation and processing circuit 61.

In the audio signal reproducing apparatus of the present embodiment, the operation unit 54 includes a storing circuit 62 in which transfer characteristics information representative of the transfer characteristics from the virtual sound sources to both ears of the listener for each predetermined angle, which is larger than that of the angular positional information of the listener calculated by the angle calculating circuit 60.

The interpolation operation and processing circuit 61 reads the information on two transfer characteristics

in the vicinity of the rotational angular position of the head represented the current angular positional information calculated by the angle calculating circuit 60 and operates the transfer characteristics in the current rotational angular position of the head by, for example, a linear interpolation processing.

The interpolation operation and processing circuit 61 may reads the information on more than two transfer characteristics in the vicinity of the current rotational angular position of the head represented by the angular positional information for performing secondary interpolation processing other than the linear interpolation processing.

The information on the transfer characteristics in the current rotational angular position obtained by the interpolation operation and processing circuit 61 is supplied to an audio signal processing circuit 63.

The audio signal processing circuit 63 is also supplied with left and right channel audio signals S_L and S_R outputted from an audio signal source 64.

The audio signal source 64 is a device for outputting predetermined left and right channel audio signals S_L and S_R and may includes, for example, various recording disc playback devices, recording the playback device of wireless receivers and the like.

The audio signal processing circuit 63 performs a signal processing which provides the left and right channel audio signals S_L and S_R fed from the audio signal source 64 with a given transfer characteristics from the virtual sound source to the both ears of the listener. The audio signal processing circuit 63 comprises first through fourth signal processing units 65a, 65b, 65c and 65d to which the transfer characteristics information in the current rotational angular positional of the head obtained by the interpolation operation and processing circuit 61. In each of signal processing units 65a, 65b, 65c and 65d, an impulse response representative of the transfer characteristics of the left and right channel audio signals S_L and S_R reproduced from a pair of left and right channel speakers which are virtual sound sources facing to a listener to each ear of the listener is preset based upon information on transfer characteristics.

In other words, the first signal processing unit 65a presents the impulse response $\{h_{RR}(t, \theta)\}$ representative of the transfer characteristics of the sound reproduced from the right channel audio signal S_R to the right ear. The second signal processing unit 65b presets the impulse response $\{h_{RL}(t, \theta)\}$ representative of the transfer characteristics of the sound reproduced from the right channel audio signal S_R to the left ear. The third signal processing unit 65c presets the impulse response $\{h_{LR}(t, \theta)\}$ representative of the transfer characteristics of the sound reproduced from the left channel audio signal S_L to the right ear. The fourth signal processing unit 65d presets the impulse response $\{h_{LL}(t, \theta)\}$ representative of the transfer characteristics of the sound reproduced from the left channel audio signal S_L to the left ear. In the audio signal processing circuit 63, the right

channel audio signal S_L is fed the first and second signal processing units 65a and 65n. In the first signal processing unit 65a, the right channel audio signal S_L is subjected to a signal processing of convolution-integration of the impulse response $\{h_{RR}(t, \theta)\}$. In the second signal processing unit 65b, the right channel audio signal S_R is subjected to a signal processing of convolution-integration of the impulse response $\{h_{RL}(t, \theta)\}$.

The left channel audio signal S_L is fed to the third and fourth signal processing units 65c and 65d. In the third signal processing unit 65c, the left channel audio signal S_L is subjected to a signal processing of convolution-integration of the impulse response $\{h_{LR}(t, \theta)\}$. In the second signal processing unit 65d, the left channel audio signal S is subjected to a signal processing of convolution-integration of the impulse response $\{h_{LL}(t, \theta)\}$.

The output signals from the first and third signal processing units 65a and 65c are applied to the right channel adder 66R and are added with each other therein. The output signal of the right channel adder 66R is fed as the right channel audio signal E_R via the right channel amplifier 68R to the right channel headphone 4R of the headphones 40 and reproduced as a sound. The output signals from the second and fourth signal processing units 64b and 64a are applied to the left channel adder 66L and are added with each other therein. The output signal of the right channel adder 66L is fed as the left channel audio signal E_R via the left channel amplifier 68L to the left channel headphone 42L of the headphone set 40 and reproduced as a sound.

In the thus formed audio signal reproducing apparatus of this embodiment, information on two transfer characteristics in the vicinity of the rotational angular position represented by the current angular positional information is read from the storing circuit 62 based upon the current angular positional information calculated by the angle calculating circuit 60. The transfer characteristics information in the current rotational angular position are operated by a linear interpolation processing in the interpolation operation circuit 61. By performing a signal processing which responds to changes in transfer characteristics in association with the movement of the listener and the rotation of the head M in real time in the audio signal processing circuit 63 based upon the transfer characteristics data, good outside head localization sensation and front localization sensation are obtained in which the virtual sound sources are not moved as similarly to the case in which an audio signal is reproduced by a pair of speaker systems which faces to the listener and are remote therefrom and with each other.

As mentioned above, in the audio signal reproducing apparatus of the present invention, information on at least two transfer characteristics in the vicinity of the rotational angular position of the head represented by the detection output from detecting means for detecting the rotational angular position of the head of the listener at a solution higher than that of the transfer characteristics information stored in the storing means is read

from the storing means. The transfer characteristics information in the rotational angular position of the head represented by the detection output are interpolation-operated by interpolation operation means. Accordingly, the amount of the information on the transfer characteristics stored in the storing means can be reduced. The audio signal processing means processes the left and right channel audio signals based upon the transfer characteristics information determined by the interpolation operation means. The processed audio signals are supplied to the headphones, resulting in that a proper binaural reproduction can be achieved for providing very natural sound image localization sensation in which the positions of the virtual sound sources do not move even if a listener moves.

Claims

1. An audio signal reproducing apparatus, comprising:

means (10, 11, 14) for detecting the rotational angular position of the head (M) of a listener corresponding to the movement of the head of the listener relative to virtual sound sources; means (22) for storing transfer characteristics information representative of transfer characteristics of direct sounds from virtual sound sources to each ear of the listener for predetermined rotational angular positions; and audio signal processing means (20, 21, 23) for forming, based upon the transfer characteristics information stored in said storing means, transfer characteristics information appropriate to the detected rotational angular position of the head and for processing left and right channel audio signals using said transfer characteristics information, whereby audio signals which have been processed by said audio signal processing means are reproduced as sounds by headphone means; characterised in that: said stored transfer characteristics information corresponds to predetermined angular positions in a quadrant of the rotational position of the head (M) of the listener and said audio signal processing means derives information for the other quadrants from the stored information.

2. An apparatus according to claim 1 in which said detecting means includes

a pair of signal detecting elements (5L, 5R) for detecting a reference signal transmitted from a reference signal source (11); means (18) for calculating the distance between said reference signal source and the head of the listener from the phase difference

between the detection output signals from said pair of the signal detecting elements and said reference signal; and
 means (19) for detecting the time difference between the detection output signals from said pair of signal detecting elements, whereby the rotational angular position of the head of the listener is calculated by using information on the distance obtained from said distance calculating means and on the time difference obtained from said time difference detecting means.

3. An apparatus according to claim 1 or 2 in which said storing means stores the transfer characteristics information representative of the transfer characteristics from said virtual sound sources to both ears of the listener in the first quadrant of the rotational angular position of the head.

4. An apparatus according to claim 1, 2 or 3, and further including control means (21) which reads from said storing means the transfer characteristics information for the angular position in the stored quadrant corresponding to the rotational angular position of the head represented, and for supplying said audio signal processing means (23) with the read transfer characteristics information together with a control signal representative of which quadrant the rotational angular position represented by the detection output from said detecting means belongs to.

5. An apparatus according to claim 1, 2, 3 or 4, in which said audio signal processing means includes

a first signal processor (28a) for applying to the right channel input audio signal a convolution-integration of the impulse response corresponding to transfer characteristics of a right channel reproduced audio signal of an input audio signal to the right ear;

a second signal processor (28b) for applying to the right channel input audio signal a convolution-integration of the impulse response corresponding to the transfer characteristics of the right channel reproduced audio signal to the left ear;

a third signal processor (28c) for applying to the left channel input audio signal a convolution-integration of the impulse response corresponding to the transfer characteristics of the left channel reproduced audio signal of the input audio signal to the right ear;

a fourth signal processor (28d) for applying to the left channel input audio signal a convolution-integration of the impulse response corresponding to the transfer characteristics of the left channel reproduced audio signal to the left

ear;

first adding means (29R) for adding the output of said first signal processing unit with the output of the third signal processing unit;

second adding means (29L) for adding the output of said second signal processing unit with the output of the fourth signal processing unit whereby, the outputs of the first and second adding means are respectively supplied to the right and left channel headphones of said headphone set (1).

6. An apparatus according to claim 5 when dependent on claim 4, wherein said audio signal processing means includes

a first selecting means (25R) for supplying said right channel input audio signal to said first and second signal processing units (28a, 28b) when the control signal represents the first or third quadrant and for supplying said right channel input audio signal to the third and fourth signal processing units (28c, 28d) when the control signal represents the second or fourth quadrant;

a second selecting means (25L) for supplying the left channel input audio signal to the third and fourth signal processing units (28c, 28d) when said control signal represent the first or third quadrant and for supplying the right channel input audio signal to the first and second signal processing units (28a, 28b) when the control signal represent the second or fourth quadrant;

a third selecting means (26R) for selecting the output of said first adding means (29R) when said control signal represents the second or fourth quadrant and for selecting the output of said second adding means (29L) when said control signal represents the second or fourth quadrant;

a fourth selecting means (26L) for selecting the output of said second adding means (29L) when the control signal represents the first or third quadrant and for selecting the output of the first adding means (29R) when the control signal represents the second or fourth quadrant;

a fifth selecting means (27R) for supplying the output of said third selecting means (26R) to the right channel headphone (2R) of the headphone set (1) when said control signal represents the first or third quadrant and for supplying the output of the third selecting means (26R) to the right channel headphone (2R) of the headphone set via a low pass filter (30R) when said control signal represents the second and fourth quadrant; and

a sixth selecting means (27L) for supplying the output of said fourth selecting means (26L) to the left channel headphone (2L) of the headphone set (1) when said control signal represents the first or third quadrant and for supplying the output of the fourth selecting means (26L) to the left channel headphone (2L) of the headphone set via a low pass filter (30L) when the control signal represents the second or fourth quadrant.

7. An apparatus according to any one of claims 1 to 5 and including interpolation processing means (61) which reads from said storing means information on transfer characteristics of at least two angles in the quadrant in the vicinity of the angle corresponding to the rotational angular position of the head and for determining transfer characteristic information corresponding to the actual rotational angular position by an interpolation processing.
8. An apparatus according to claim 7 in which said interpolation processing means (61) supplies transfer characteristics information determined by said interpolation processing to said audio signal processing means (63) together with a control signal representing which quadrant the actual rotational angular position of the head represented by the detection output from said detecting means belongs to.

Patentansprüche

1. Audiosignal-Wiedergabevorrichtung mit:

einer Einrichtung (10, 11, 14) zur Erfassung der Drehwinkelstellung des Kopfes (M) eines Hörers entsprechend der Bewegung des Kopfes des Hörers bezüglich virtueller Klangquellen; einer Einrichtung (22) zur Speicherung von Übertragungskennlinien-Information, die Übertragungskennlinien von Direktklang von virtuellen Klangquellen zu jedem Ohr des Hörers für vorbestimmte Drehwinkelstellungen wiedergibt; und einer Audiosignal-Verarbeitungseinrichtung (20, 21, 23) zur Bildung einer für die erfaßte Drehwinkelstellung des Kopfes geeigneten Übertragungskennlinien-Information auf Grundlage der in der Speichereinrichtung gespeicherten Übertragungskennlinien-Information, und zur Verarbeitung von Audiosignalen eines linken und eines rechten Kanales unter Verwendung der Übertragungskennlinien-Information, wobei durch die Audiosignal-Verarbeitungseinrichtung verarbeitete Audiosignale als Klang mittels einer Kopfhörereinrichtung wiedergegeben werden,

dadurch gekennzeichnet,

daß die gespeicherte Übertragungskennlinien-Information vorbestimmten Winkelstellungen in einem Quadranten der Drehwinkelstellung des Kopfes (M) des Hörers entspricht, und die Audiosignal-Verarbeitungseinrichtung eine Information für die anderen Quadranten aus der gespeicherten Information ableitet.

2. Vorrichtung nach Anspruch 1,

bei der die Erfassungseinrichtung aufweist: ein Paar von Signalerfassungselementen (5L, 5R) zur Erfassung eines Referenzsignales, das von einer Referenzsignalquelle (11) übertragen wird; eine Einrichtung (18) zur Berechnung des Abstandes zwischen der Referenzsignalquelle und dem Kopf des Hörers aus der Phasendifferenz zwischen dem Erfassungs-Ausgangssignalen von dem Paar von Signalerfassungselementen und dem Referenzsignal; und eine Einrichtung (19) zur Erfassung der Zeitdifferenz zwischen Erfassungs-Ausgangssignalen von dem Paar von Signalerfassungselementen, wobei die Drehwinkelposition des Kopfes des Hörers unter Verwendung der Information bezüglich des Abstandes, die von der Abstandsberechnungseinrichtung erhalten wird, und abhängig von der Zeitdifferenz berechnet wird, die von der Zeitdifferenz-Erfassungseinrichtung erhalten wird.

3. Vorrichtung nach Anspruch 1 oder 2,

bei der die Speichereinrichtung die Übertragungskennlinien-Information speichert, die die Übertragungskennlinien von den virtuellen Klangquellen zu beiden Ohren des Hörers in dem ersten Quadranten der Drehwinkelstellung des Kopfes wiedergibt.

4. Vorrichtung nach einem der Ansprüche 1, 2 oder 3,

die weiterhin eine Steuereinrichtung (21) aufweist, die aus der Speichereinrichtung die Übertragungskennlinien-Information der Winkelstellung in dem gespeicherten Quadranten ausliest, die der dargestellten Drehwinkelposition des Kopfes entspricht, und zur Weitergabe der ausgelesenen Übertragungskennlinien-Information zu der Audiosignal-Verarbeitungseinrichtung (23) zusammen mit einem Steuerungssignal, das anzeigt, zu welchem Quadranten die Drehwinkelposition gehört, die durch das Erfassungs-Ausgangssignal von der Erfassungseinrichtung wiedergegeben wird.

5. Vorrichtung nach einem der Ansprüche 1 bis 4,

bei der Audiosignal-Verarbeitungseinrichtung aufweist:

eine erste Signalverarbeitungseinrichtung (28a) zur Ausübung auf das Eingangs-Audiosignal des rechten Kanales einer Faltung der Impulsantwort entsprechend einer Übertragungskennlinie eines wiedergegebenen rechten Audiosignales eines Eingangs-Audiosignales zu dem rechten Ohr; 5
eine zweite Signalverarbeitungseinrichtung (28b) zur Ausübung auf das Eingangs-Audiosignal des rechten Kanales einer Faltung der Impulsantwort entsprechend der Übertragungskennlinien des rechten wiedergegebenen Audiosignales zu dem linken Ohr; eine dritte Signalverarbeitungseinrichtung (28c) zur Ausübung auf das Eingangs-Audiosignal eines linken Kanales einer Faltung der Impulsantwort entsprechend der Übertragungskennlinien des wiedergegebenen Audiosignales des linken Kanales des Eingangs-Audiosignales zu dem rechten Ohr; 10
eine vierte Signalverarbeitungseinrichtung (28d) zur Ausübung auf das Eingangs-Audiosignal des linken Kanales einer Faltung der Impulsantwort entsprechend der Übertragungskennlinien des wiedergegebenen Audiosignales des linken Kanales zu dem linken Ohr; 15
eine erste Addiereinrichtung (29R) zum Addieren des Ausgangssignales von der ersten Signalverarbeitungseinrichtung zu dem Ausgangssignal der dritten Signalverarbeitungseinrichtung; und 20
eine zweite Addiereinrichtung (29L) zur Addition des Ausgangssignales der zweiten Signalverarbeitungseinrichtung zu dem Ausgangssignal der vierten Signalverarbeitungseinrichtung, wobei die Ausgangssignale der ersten und zweiten Addiereinrichtung zu den Kopfhörern des rechten bzw. linken Kanales der Kopfhöreranordnung (1) gegeben werden. 25

6. Vorrichtung nach Anspruch 5, wenn zurückbezogen auf Anspruch 4, 30

wobei die Audiosignal-Verarbeitungseinrichtung aufweist:

eine erste Wahleinrichtung (25R) zur Versorgung der ersten und zweiten Signalverarbeitungseinheit (28a, 28b) mit dem rechten Kanal-Eingangs-Audiosignal, wenn das Steuersignal den ersten oder dritten Quadranten wiedergibt, und zur Versorgung der dritten und vierten Signalverarbeitungseinheit (28c, 28d) mit dem rechten Kanal-Eingangs-Audiosignal, wenn das Steuersignal den zweiten oder vierten 35
Quadranten wiedergibt; 40
eine zweite Wahleinrichtung (25L) zur Versorgung der ersten und zweiten Signalverarbeitungseinheit (28a, 28b) mit dem linken Kanal-Eingangs-Audiosignal, wenn das Steuersignal den ersten oder dritten Quadranten anzeigt, und zur Versorgung der dritten und vierten Signalverarbeitungseinheit (28c, 28d) mit dem linken Kanal-Eingangs-Audiosignal, wenn das Steuersignal den zweiten oder vierten Quadranten wiedergibt; 45
eine dritte Wahleinrichtung (26R) zur Wahl des Ausgangssignales der ersten Addiereinrichtung (29R), wenn das Steuersignal den ersten oder dritten Quadranten wiedergibt, und zur Wahl des Ausgangssignales der zweiten Addiereinrichtung (29L), wenn das Steuersignal den zweiten oder vierten Quadranten wiedergibt; 50
eine vierte Wahleinrichtung (26L) zur Wahl des Ausgangssignales der zweiten Addiereinrichtung (29L), wenn das Steuersignal den ersten oder dritten Quadranten wiedergibt, und zur Wahl des Ausgangssignales der ersten Addiereinrichtung (29R), wenn das Steuersignal den zweiten oder vierten Quadranten wiedergibt; 55
eine fünfte Wahleinrichtung (27R) zur Versorgung des rechten Kopfhörers (2R) der Kopfhöreranordnung mit dem Ausgangssignal der dritten Wahleinrichtung (26R), wenn das Steuersignal den ersten oder dritten Quadranten wiedergibt, und zur Versorgung des rechten Kopfhörers (2R) der Kopfhöreranordnung mit dem Ausgangssignal der dritten Wahleinrichtung (26R) mittels eines Tiefpaßfilters (30R), wenn das Steuersignal den zweiten und vierten Quadranten anzeigt; und
eine sechste Wahleinrichtung (27L) zur Versorgung des linken Kopfhörers (2L) der Kopfhöreranordnung (1) mit dem Ausgangssignal der vierten Wahleinrichtung (26L), wenn das Steuersignal den ersten oder dritten Quadranten angibt, und zur Versorgung des linken Kopfhörers (2L) der Kopfhöreranordnung mit dem Ausgangssignal der vierten Wahleinrichtung (26L) mittels eines Tiefpaßfilters (30L), wenn das Steuersignal den zweiten oder vierten Quadranten wiedergibt.

7. Vorrichtung nach einem der Ansprüche 1 bis 5,

die eine Interpolations-Verarbeitungseinrichtung (61) aufweist, die aus der Speichereinrichtung eine Information bezüglich Übertragungskennlinien von wenigstens zwei Winkeln in dem Quadranten in der Nähe des Winkels ausliest, der der Drehwinkelstellung des Kopfes ent-

spricht, und zur Festlegung von der Übertragungskennlinien-Information, die der aktuellen Drehwinkelposition entspricht, durch eine Interpolationsverarbeitung.

8. Vorrichtung nach Anspruch 7,

bei der die Interpolations-Verarbeitungseinrichtung (61) eine Übertragungskennlinien-Information, die durch die Interpolationsverarbeitung bestimmt wird, zu der Audiosignal-Verarbeitungseinrichtung (63) zusammen mit einem Steuersignal gibt, das anzeigt, in welchem Quadranten die aktuelle Rotationswinkelstellung des Kopfes liegt, die durch das Erfassungsausgangssignal von der Erfassungseinrichtung wiedergegeben wird.

Revendications

1. Appareil de reproduction de signal audio comprenant :

un moyen (10, 11, 14) pour détecter la position angulaire en rotation de la tête (M) d'un auditeur correspondant au déplacement de la tête de l'auditeur par rapport à des sources de son virtuelles ;

un moyen (22) pour stocker une information de caractéristique de transfert représentative d'une caractéristique de transfert de sons directs provenant de sources de son virtuelles pour chaque oreille de l'auditeur pour des positions angulaires en rotation prédéterminées ; et

un moyen de traitement de signal audio (20, 21, 23) pour former, sur la base de l'information de caractéristique de transfert stockée dans ledit moyen de stockage, une information de caractéristique de transfert qui convient pour la position angulaire en rotation détectée de la tête et pour traiter des signaux audio de canaux gauche et droit en utilisant ladite information de caractéristique de transfert et ainsi, des signaux audio qui ont été traités par ledit moyen de traitement de signal audio sont reproduits en tant que sons par un moyen d'écouteur, caractérisé en ce que :

ladite information de caractéristique de transfert stockée correspond à des positions angulaires prédéterminées dans un quadrant de la position en rotation de la tête (M) de l'auditeur et ledit moyen de traitement de signal audio délivre une information pour les autres quadrants à partir de l'information stockée.

2. Appareil selon la revendication 1, dans lequel ledit

moyen de détection inclut :

une paire d'éléments de détection de signal (5L, 5R) pour détecter un signal de référence émis depuis une source de signal de référence (11) ;

un moyen (18) pour calculer la distance entre ladite source de signal de référence et la tête de l'auditeur à partir de la différence de phase entre les signaux de sortie de détection provenant de ladite paire d'éléments de détection de signal et ledit signal de référence ; et

un moyen (19) pour détecter la différence temporelle entre les signaux de sortie de détection provenant de ladite paire d'éléments de détection de signal et ainsi, la position angulaire en rotation de la tête de l'auditeur est calculée en utilisant une information concernant la distance obtenue à partir dudit moyen de calcul de distance et concernant la différence temporelle obtenue à partir dudit moyen de détection de différence.

3. Appareil selon la revendication 1 ou 2, dans lequel ledit moyen de stockage stocke l'information de caractéristique de transfert représentative de la caractéristique de transfert provenant desdites sources de son virtuelles pour les deux oreilles de l'auditeur dans le premier quadrant de la position angulaire en rotation de la tête.

4. Appareil selon la revendication 1, 2 ou 3, incluant en outre un moyen de commande (21) qui lit dans ledit moyen de stockage l'information de caractéristique de transfert pour la position angulaire dans le quadrant stocké correspondant à la position angulaire en rotation de la tête représentée, et pour appliquer audit moyen de traitement de signal audio (23) l'information de caractéristique de transfert lue en association avec un signal de commande représentatif du quadrant auquel la position angulaire en rotation représentée par la sortie de détection provenant dudit moyen de détection appartient.

5. Appareil selon la revendication 1, 2, 3 ou 4, dans lequel ledit moyen de traitement de signal audio inclut :

un premier processeur de signal (28a) pour appliquer sur le signal audio d'entrée de canal droit une convolution intégration de la réponse impulsionnelle correspondant à une caractéristique de transfert d'un signal audio reproduit de canal droit d'un signal audio d'entrée appliqué sur l'oreille droite ;

un second processeur de signal (28b) pour appliquer sur le signal audio d'entrée de canal droit une convolution-intégration de la réponse

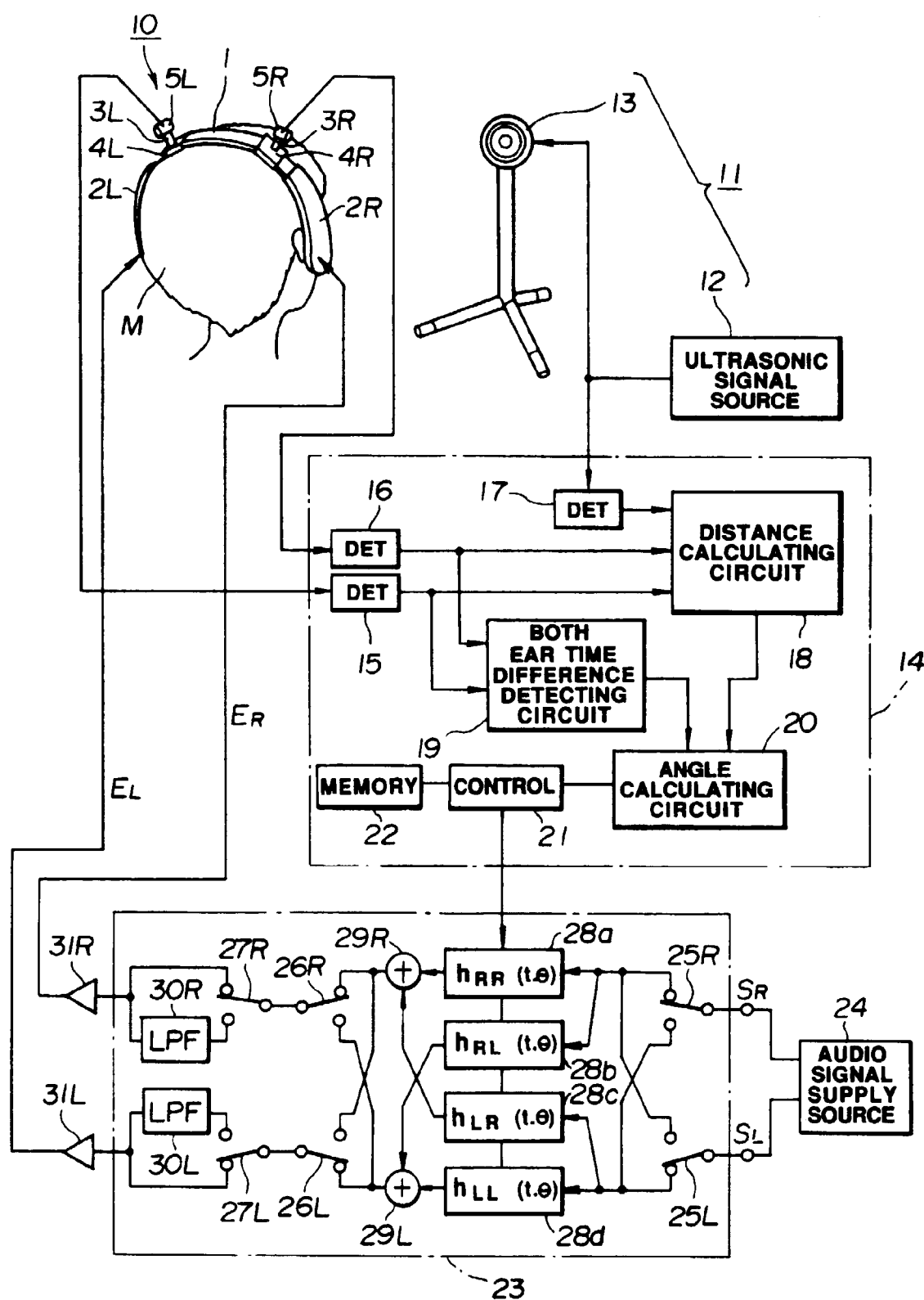
impulsionnelle correspondant à la caractéristique de transfert du signal audio reproduit de canal droit appliqué sur l'oreille gauche ;
 un troisième processeur de signal (28c) pour appliquer sur le signal audio d'entrée de canal gauche une convolution-intégration de la réponse impulsionnelle correspondant à la caractéristique de transfert du signal audio reproduit de canal gauche du signal audio d'entrée appliqué sur l'oreille droite ;
 un quatrième processeur de signal (28d) pour appliquer sur le signal audio d'entrée de canal gauche une convolution-intégration de la réponse impulsionnelle correspondant à la caractéristique de transfert du signal audio reproduit de canal gauche appliqué sur l'oreille gauche ;
 un premier moyen d'addition (29R) pour additionner la sortie de ladite première unité de traitement de signal à la sortie de la troisième unité de traitement de signal ;
 un second moyen d'addition (29L) pour additionner la sortie de ladite seconde unité de traitement de signal à la sortie de la quatrième unité de traitement de signal et ainsi, les sorties des premier et second moyens d'addition sont respectivement appliquées sur les écouteurs de canaux droit et gauche dudit jeu d'écouteurs (1).

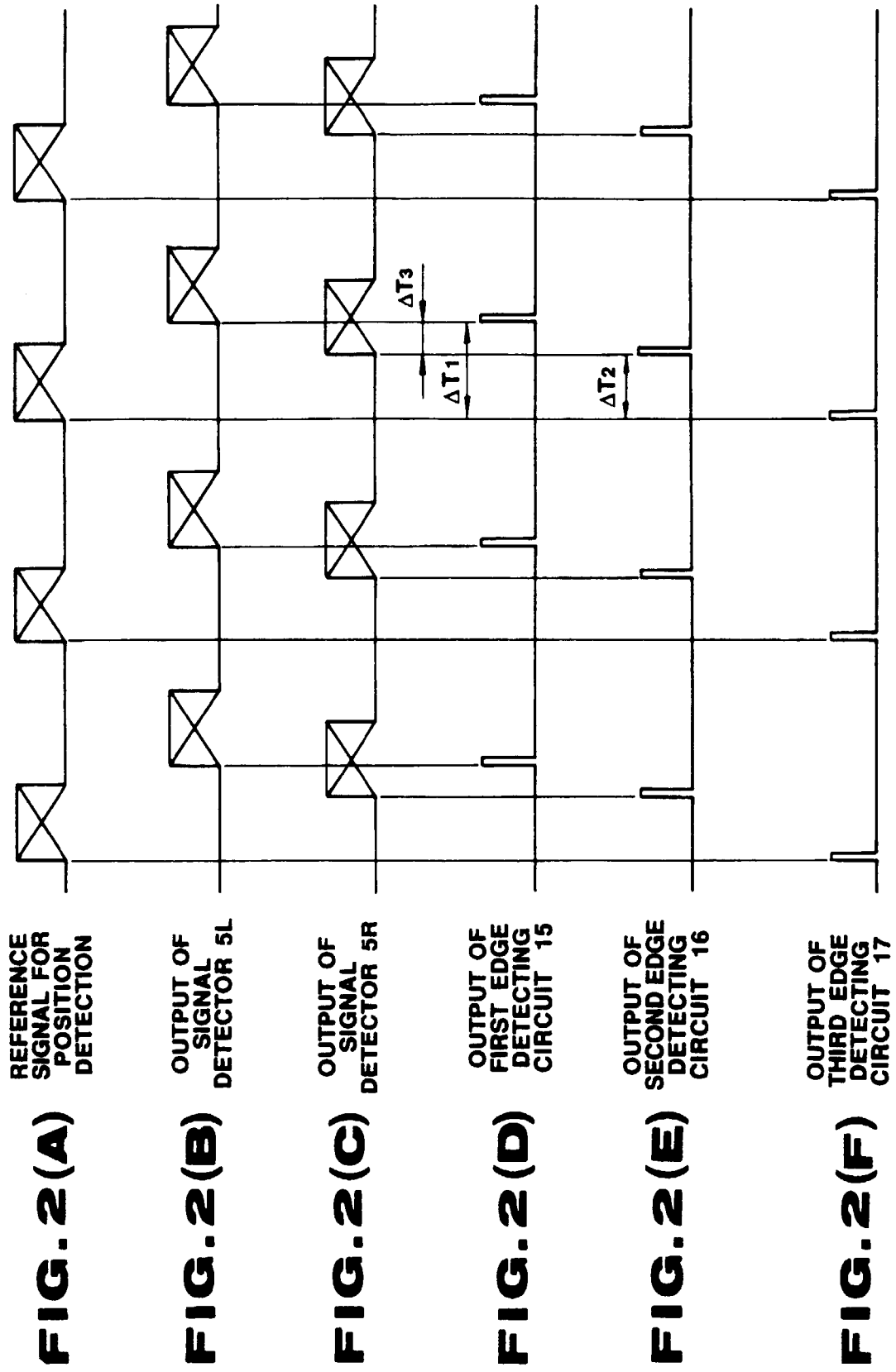
6. Appareil selon la revendication 5 lorsqu'elle dépend de la revendication 4, dans lequel ledit moyen de traitement de signal audio inclut:

un premier moyen de sélection (25R) pour appliquer ledit signal audio d'entrée de canal droit sur lesdites première et seconde unités de traitement de signal (28a, 28b) lorsque le signal de commande représente le premier ou troisième quadrant et pour appliquer ledit signal audio d'entrée de canal droit sur les troisième et quatrième unités de traitement de signal (28c, 28d) lorsque le signal de commande représente le second ou quatrième quadrant ;
 un second moyen de sélection (25L) pour appliquer le signal audio d'entrée de canal gauche sur les troisième et quatrième unités de traitement de signal (28c, 28d) lorsque ledit signal de commande représente le premier ou troisième quadrant et pour appliquer le signal audio d'entrée de canal droit sur les première et seconde unités de traitement de signal (28a, 28b) lorsque le signal de commande représente le second ou quatrième quadrant ;
 un troisième moyen de sélection (26R) pour sélectionner la sortie dudit premier moyen d'addition (29R) lorsque ledit signal de commande représente le second ou quatrième quadrant et pour sélectionner la sortie dudit second moyen

d'addition (29L) lorsque ledit signal de commande représente le second ou quatrième quadrant ;
 un quatrième moyen de sélection (26L) pour sélectionner la sortie dudit second moyen d'addition (29L) lorsque le signal de commande représente le premier ou troisième quadrant et pour sélectionner la sortie du premier moyen d'addition (29R) lorsque le signal de commande représente le second ou quatrième quadrant ;
 un cinquième moyen de sélection (27R) pour appliquer la sortie dudit troisième moyen de sélection (26R) sur l'écouteur de canal droit (2R) du jeu d'écouteurs (1) lorsque ledit signal de commande représente le premier ou troisième quadrant et pour appliquer la sortie du troisième moyen de sélection (26R) sur l'écouteur de canal droit (2R) du jeu d'écouteurs via un filtre passe-bas (30R) lorsque ledit signal de commande représente le second ou quatrième quadrant; et
 un sixième moyen de sélection (27L) pour appliquer la sortie dudit quatrième moyen de sélection (26L) sur l'écouteur de canal gauche (2L) du jeu d'écouteurs (1) lorsque ledit signal de commande représente le premier ou troisième quadrant et pour appliquer la sortie du quatrième moyen de sélection (26L) sur l'écouteur de canal gauche (2L) du jeu d'écouteurs via un filtre passe-bas (30L) lorsque le signal de commande représente le second ou quatrième quadrant.

7. Appareil selon l'une quelconque des revendications 1 à 5 et incluant un moyen de traitement d'interpolation (61) qui lit dans ledit moyen de stockage une information concernant une caractéristique de transfert d'au moins deux angles dans le quadrant au voisinage de l'angle correspondant à la position angulaire en rotation de la tête et pour déterminer une information de caractéristique de transfert correspondant à la position angulaire en rotation réelle à l'aide d'un traitement d'interpolation.
8. Appareil selon la revendication 7, dans lequel ledit moyen de traitement d'interpolation (61) applique une information de caractéristique de transfert déterminée par ledit traitement d'interpolation sur ledit moyen de traitement de signal audio (63) en association avec un signal de commande représentant à quel quadrant la position angulaire en rotation réelle de la tête représentée par la sortie de détection provenant dudit moyen de détection appartient.

**FIG. 1**



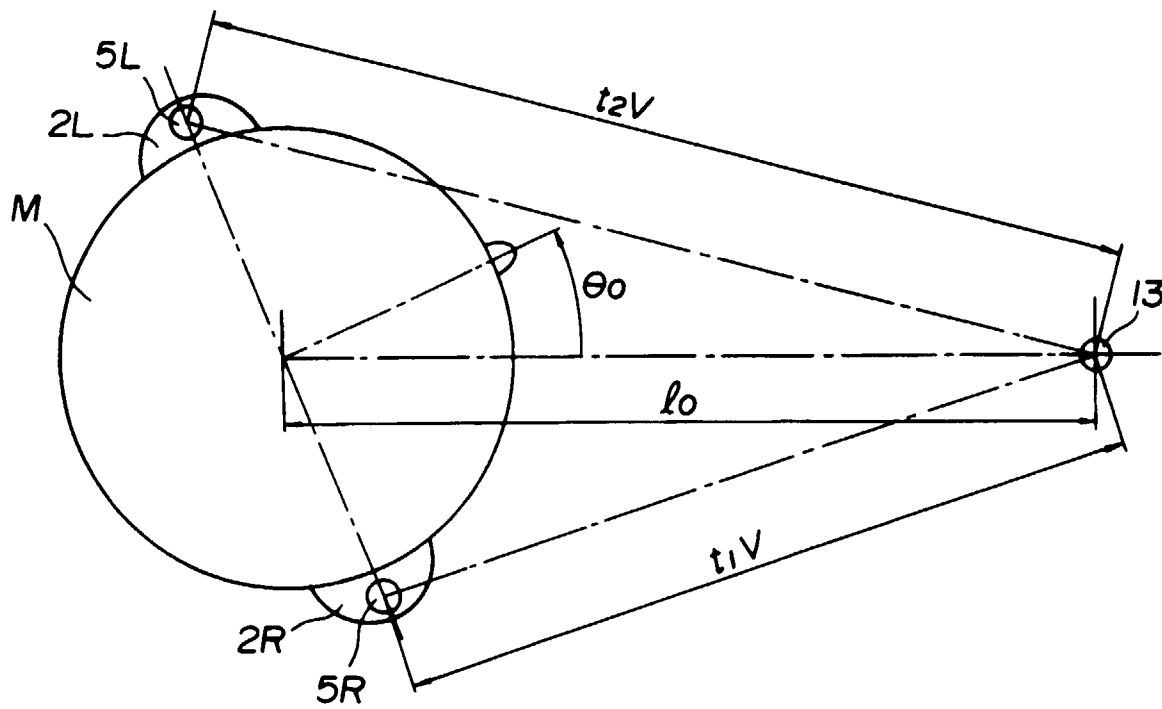


FIG. 3

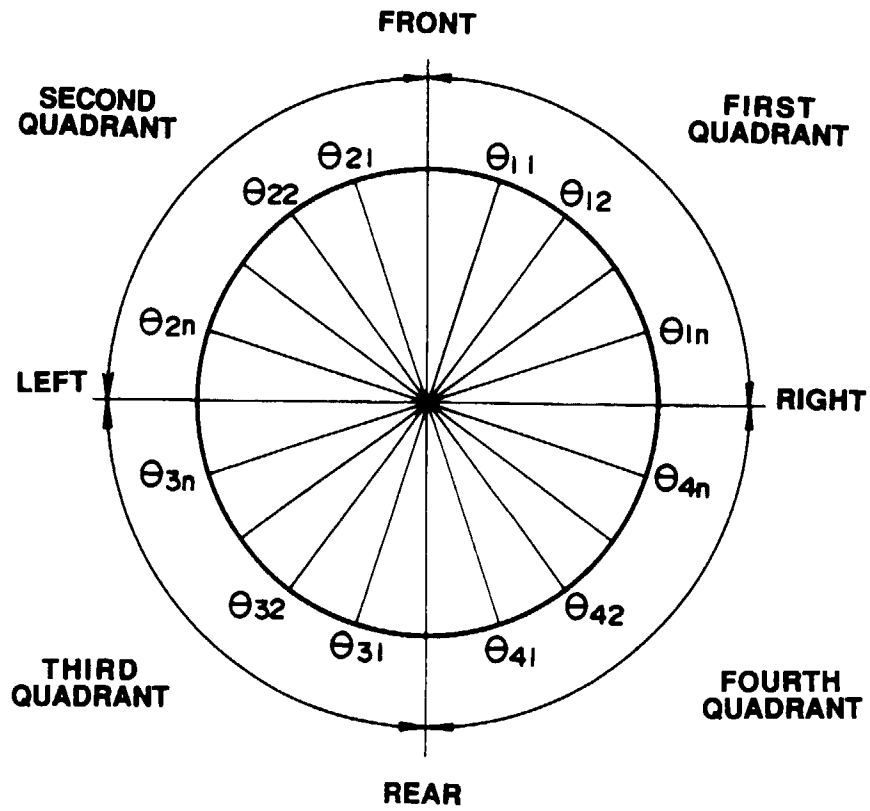


FIG. 4

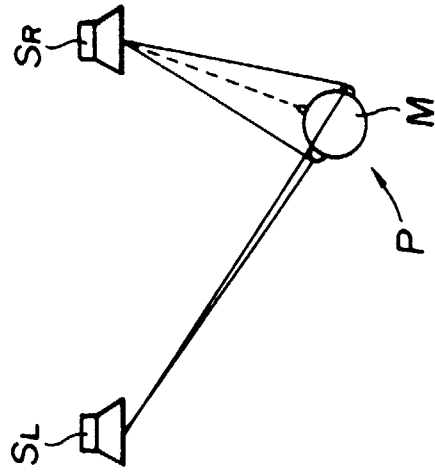


FIG. 5(C)

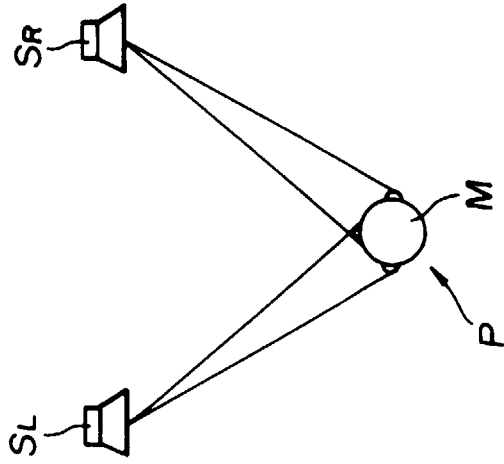


FIG. 5(B)

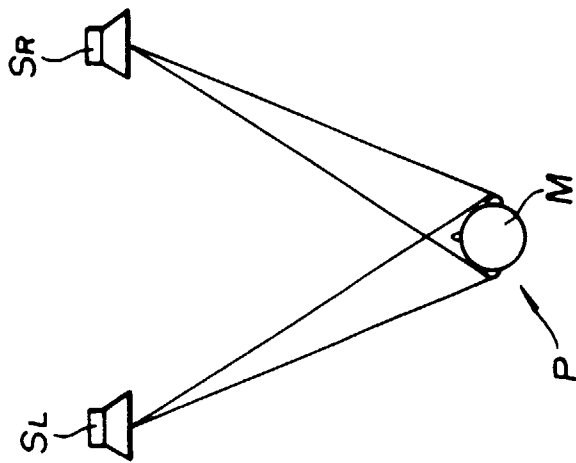


FIG. 5(A)

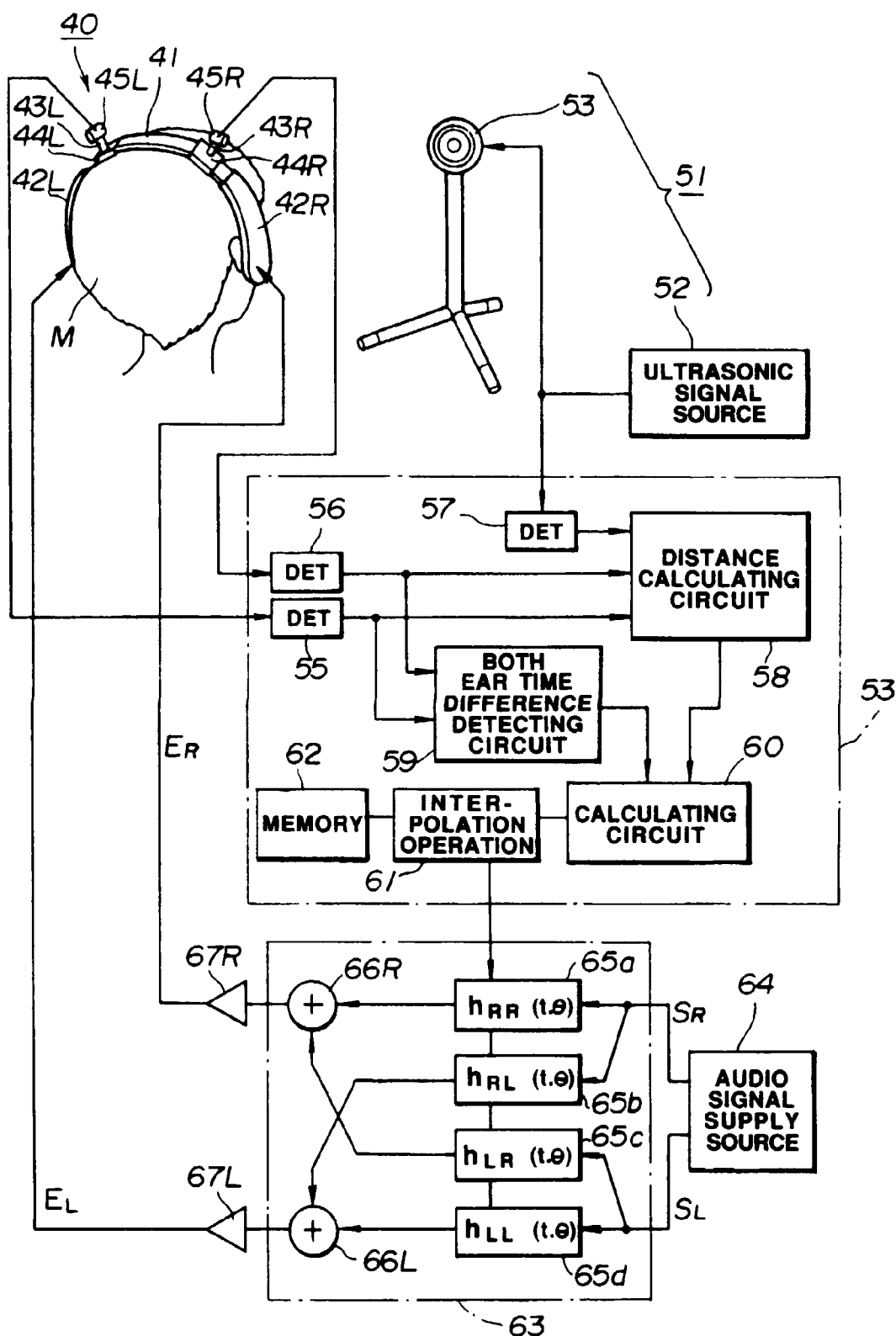


FIG. 6