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(54) **APPARATUS AND METHOD FOR PROCESSING AUDIO SIGNALS RECORDED ON A MEDIUM**

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(73) Assignee: **LG Electronics Inc.**, Seoul (KR)

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(\*) Notice: This patent is subject to a terminal disclaimer.

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**Related U.S. Patent Documents**

Reissue of:

(64) Patent No.: **6,721,249**  
Issued: **Apr. 13, 2004**  
Appl. No.: **09/209,280**  
Filed: **Dec. 11, 1998**

(57) **ABSTRACT**

[This invention relates to a karaoke CD format and a device for controlling audio signals utilizing the karaoke CD format which provides a user with selective reproduction of the voice of a player. The karaoke CD format includes a CD format divided into video sectors and audio sectors. The audio sectors are further divided into a part for recording accompaniment sound and playing sound at the same time and a part for recording accompaniment sound only. There is provided record coding information for distinguishing the parts.] *An apparatus, method and storage medium for reproducing digital data, are discussed. According to an embodiment, the apparatus includes a demultiplexing circuit to receive data units of digital audio data and separate a system time information from at least one data unit of digital audio data, the digital audio data including at least two of: 1) a first type of digital audio data, and 2) a second type of digital audio data which includes at least one content different from the first type of digital audio data; each type of digital audio data being specified by attribute information, which has: a) first information for indicating a coding mode of the digital audio data, and b) second information, different than the first information, for identifying one of the first type of digital audio data and the second type of digital audio data.*

U.S. Applications:

(63) Continuation of application No. 11/058,630, filed on Feb. 16, 2005, which is a continuation of application No. 10/848,616, filed on May 19, 2004, now abandoned, which is a division of application No. 08/735,572, filed on Oct. 23, 1996, now Pat. No. 5,535,467, which is a continuation of application No. 08/307,288, filed on Sep. 16, 1994, now abandoned.

(30) **Foreign Application Priority Data**

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(51) **Int. Cl.**  
**G11B 7/005** (2006.01)

(52) **U.S. Cl.** ..... **369/47.16; 369/47.2; 369/47.23; 369/53.37**

(58) **Field of Classification Search** ..... None  
See application file for complete search history.

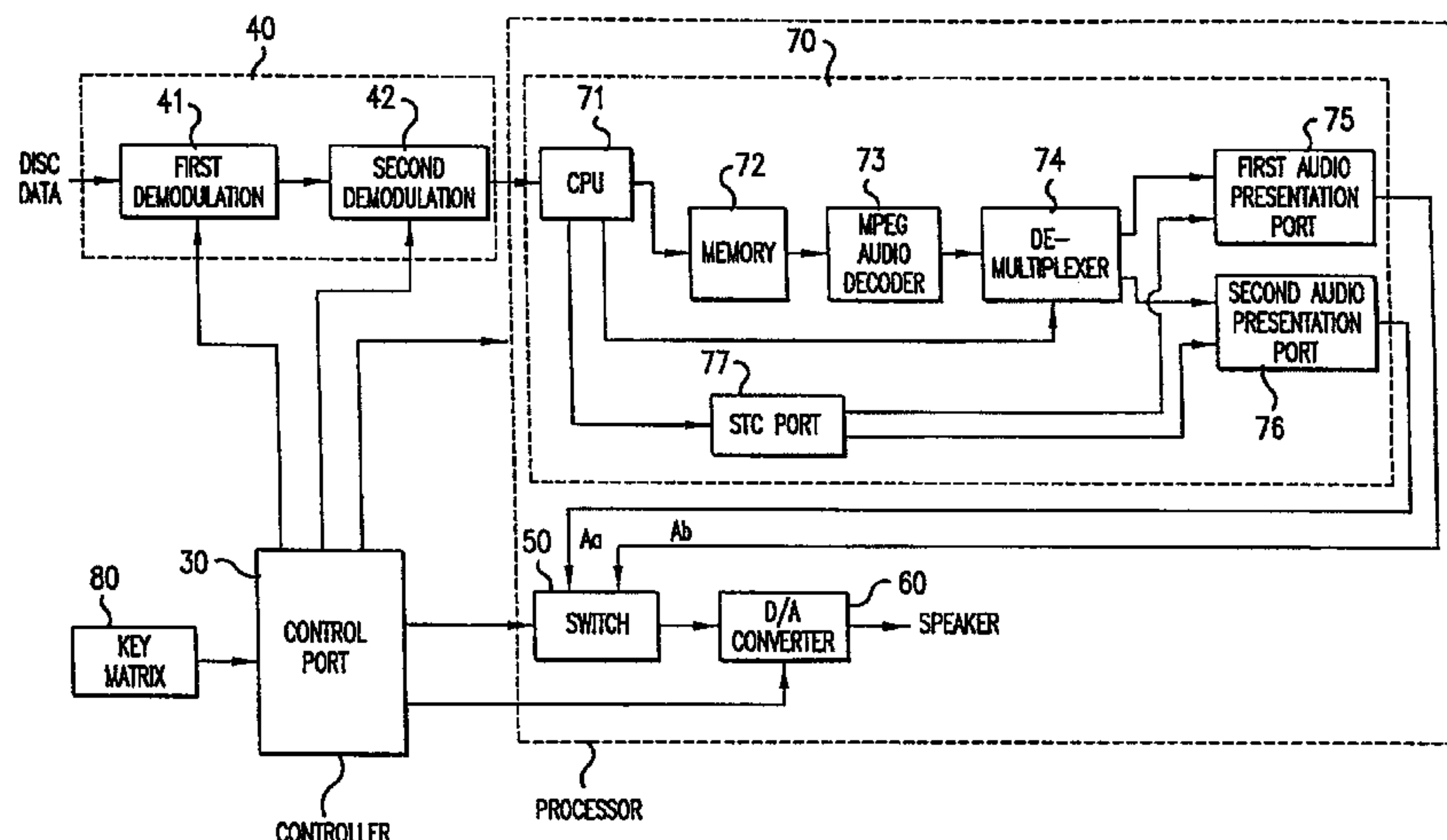
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**17 Claims, 6 Drawing Sheets**



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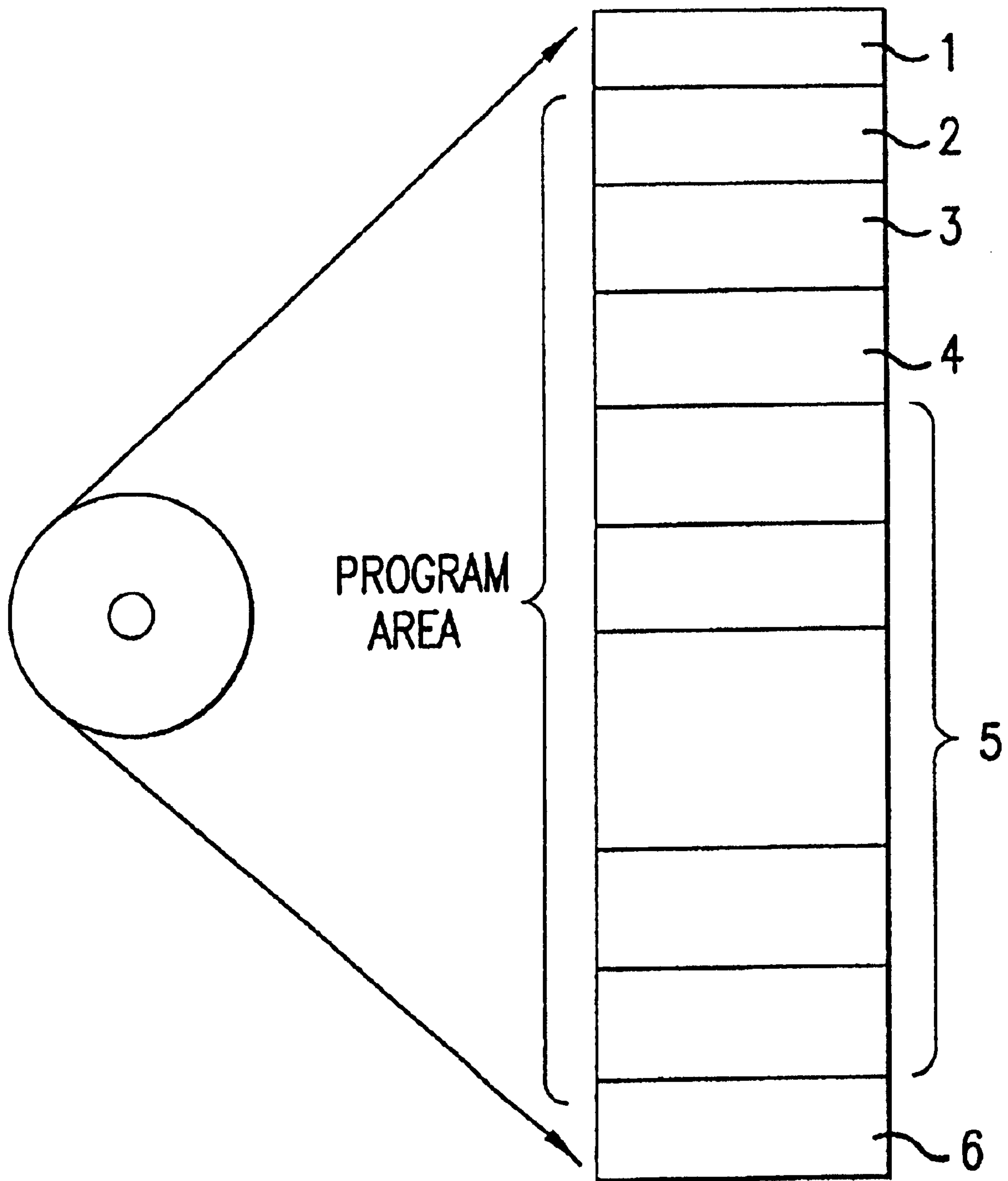


FIG. 1  
PRIOR ART

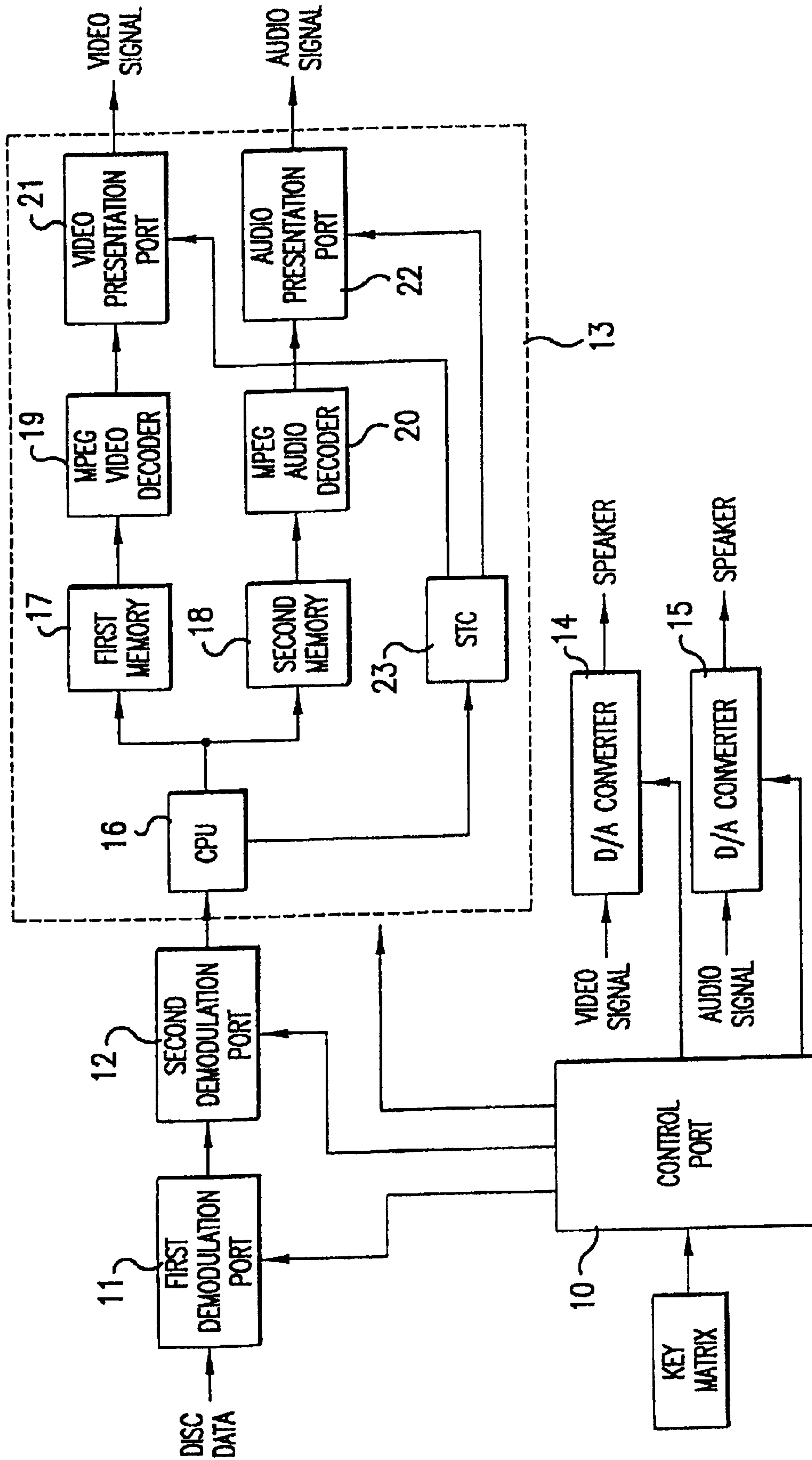


FIG. 2  
PRIOR ART

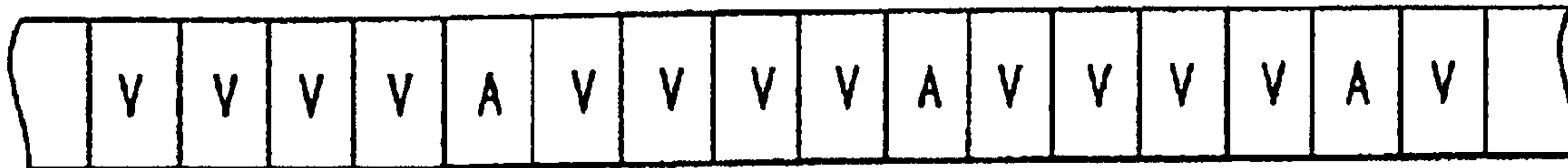


FIG.3  
PRIOR ART

	MPEG VIDEO	MPEG AUDIO
FILE NUMBER	SEQUENCE NUMBER	SEQUENCE NUMBER
CHANNEL NUMBER	\$ 01	\$ 01
SUBMODE	%x11x001x	%x11x010x
CODING NUMBER	\$ 0F	\$ 7F

FIG.4

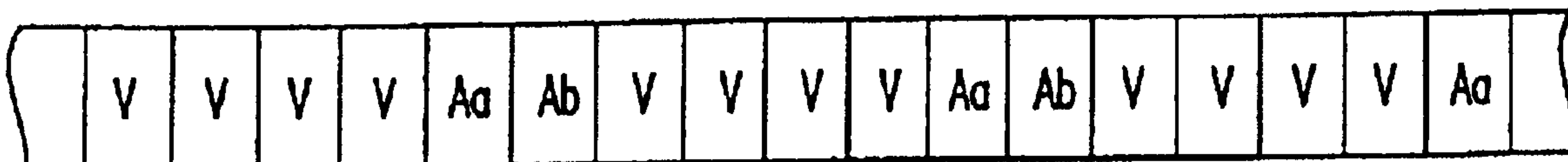


FIG.5



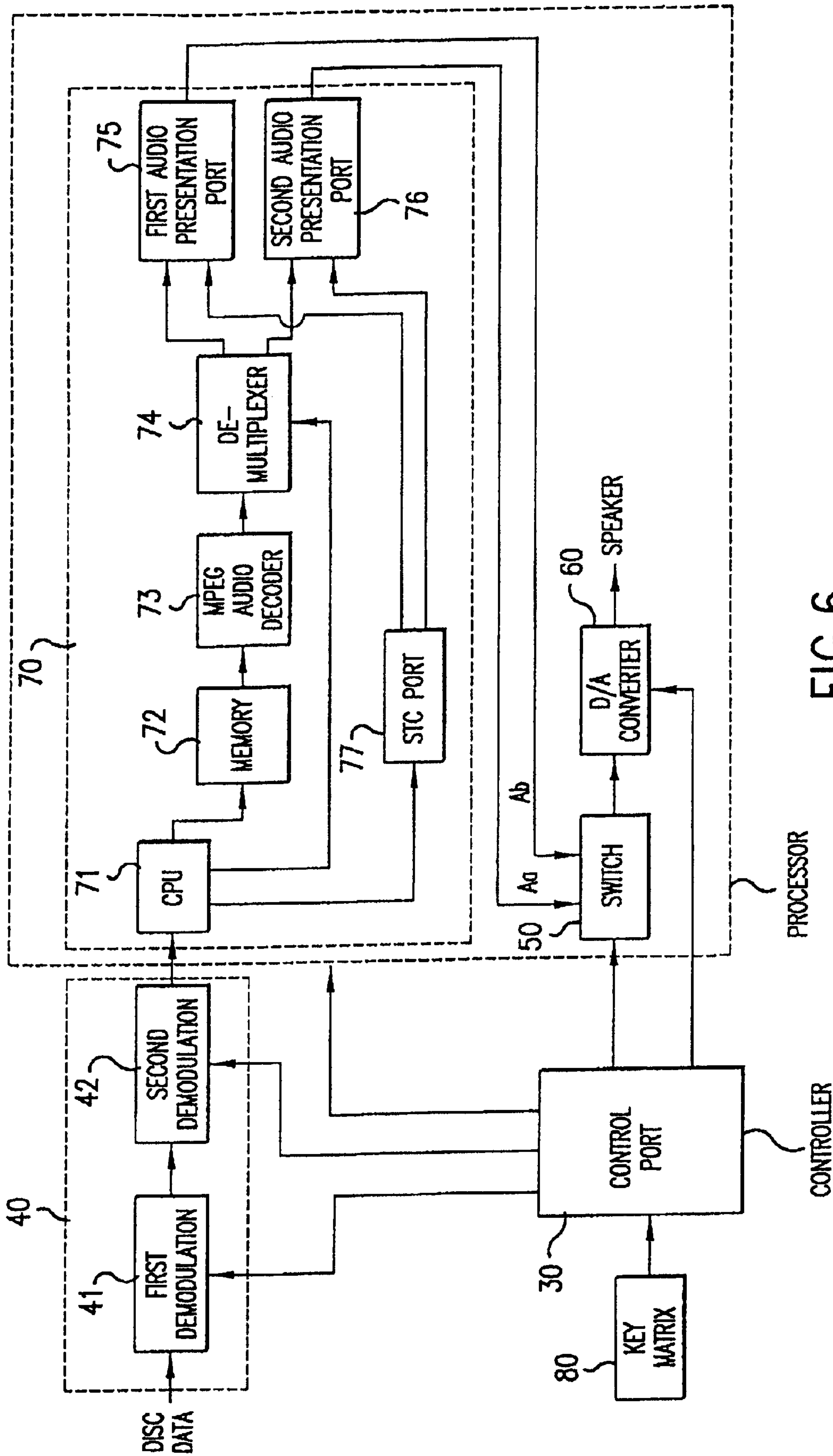


FIG. 6

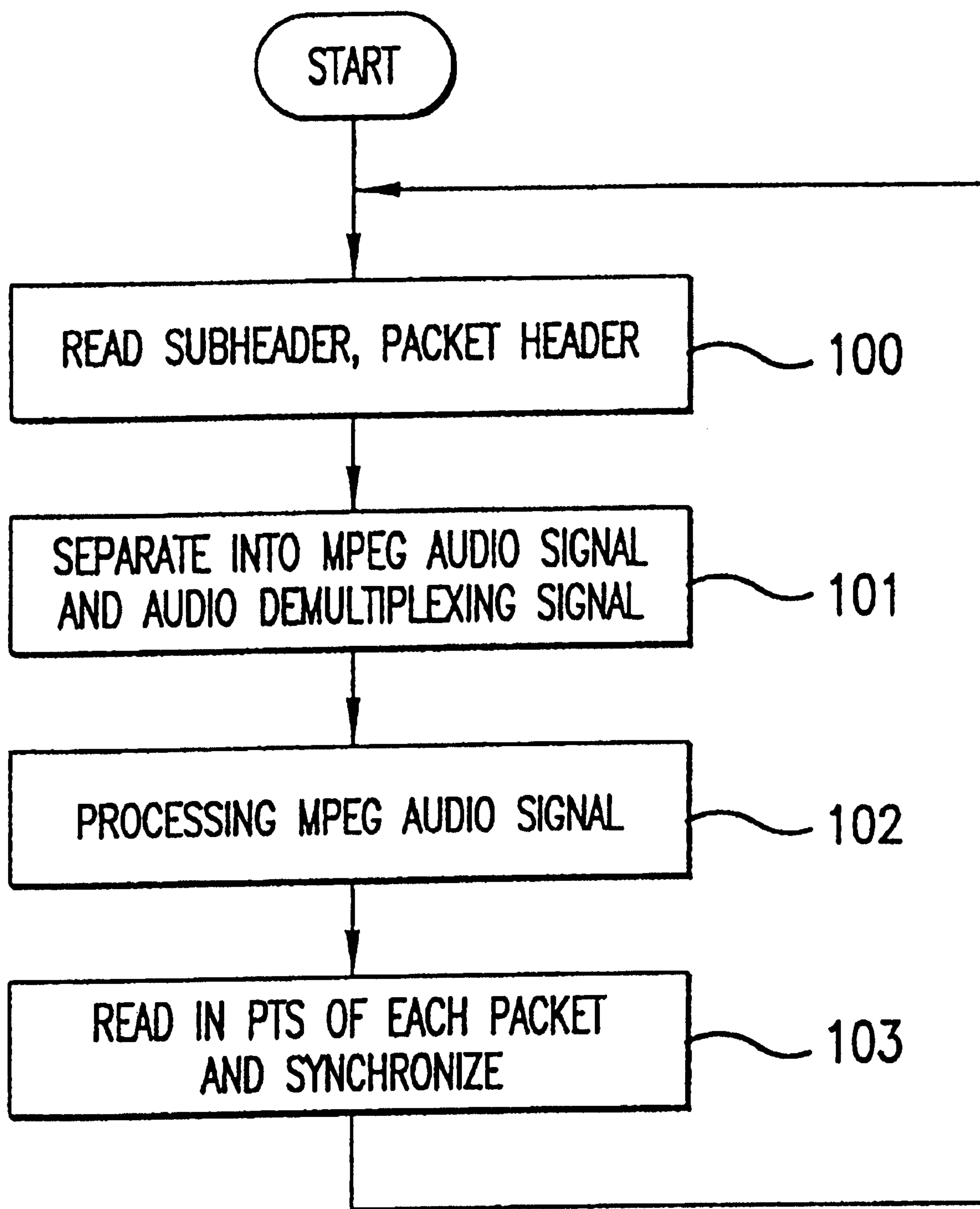


FIG.7

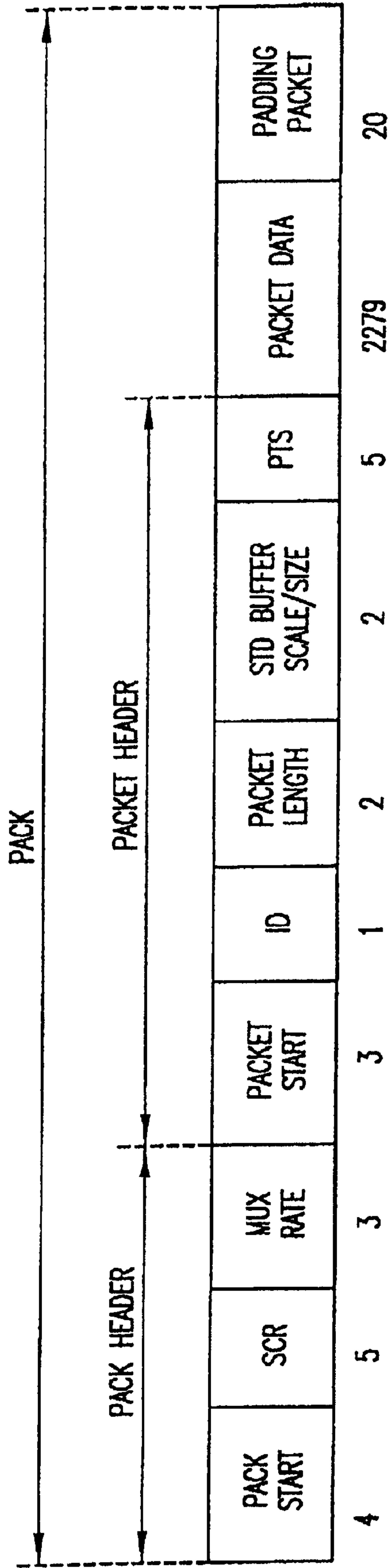


FIG.8



**APPARATUS AND METHOD FOR  
PROCESSING AUDIO SIGNALS RECORDED  
ON A MEDIUM**

**Matter enclosed in heavy brackets [ ] appears in the original patent but forms no part of this reissue specification; matter printed in italics indicates the additions made by reissue.**

[This is a division of application Ser. No. 08/735,572, filed Oct. 23, 1996, now U.S. Pat. No. 6,535,467, which is a continuation of 08/307,288, filed Sep. 16, 1994, now abandoned.—all of which are incorporated herein by reference.] *This application is a Continuation Reissue Application of U.S. Reissue application Ser. No. 11/058,630 filed on Feb. 16, 2005, which is Continuation reissue of U.S. Reissue application Ser. No. 10/848,616 filed on May 19, 2000 (now abandoned), which is a reissue of U.S. Pat. No. 6,721,249 (application Ser. No. 09/209,280 filed Dec. 11, 1998), which is a divisional of U.S. application Ser. No. 08/735,572 filed Oct. 23, 1996 (U.S. Pat. No. 6,535,647), which is a continuation of U.S. application Ser. No. 08/307,288 filed Sep. 16, 1994 (now abandoned), all of which are herein incorporated by reference. The present application also claims foreign priority benefits of Korean Patent Application No. 93-18841 filed Sep. 17, 1993 in Republic of Korea. The entirety of each of the above-identified documents is hereby incorporated by reference. Note: More than one reissue application has been filed for the reissue of U.S. Pat. No. 6,721,249. The reissue applications are application Ser. Nos. 10/848,616 (grandparent of the present application—now abandoned); 10/848,617 (child of U.S. Pat. No. 6,721,249—now abandoned); 11/058,630 (parent of the present application); 11/905,506 (child of application Ser. No. 11/058,630); 12/142,642; 12/142,658; 12/142,663; 12/142,674; 12/142,680 (the present application); and 12/142,684.*

**BACKGROUND OF THE INVENTION**

**1. Field of the Invention**

This invention relates to a karaoke CD system, more particularly to a karaoke CD format and a device for processing audio signals utilizing the karaoke CD format which provides a user with selective reproduction of the voice of a player.

**2. Description of Related Art**

As shown in FIG. 1, a general disc data layout of a karaoke CD includes lead in/lead out areas **1** and **6** formed at the lead in and lead out of the CD for encoding CD-ROMXA data containing empty sectors of mode **2**/form **2** therein, and program areas **2**, **3**, **4**, and **5**.

The program areas include a volume descriptor area **2** formed next to the lead in area **1** containing the path name of a CDI application program described in compact disc bridge system description. The basic data area **3** is for recording the title of a song, the name of the singer, the composer of the song, and the songwriter. The CDI application program area **4** is for recording an application program required for converting into compact disc bridge system description which will facilitate the operation of a disc in CD-RTOS (Real Time Operating System) in a CD-I/FMV player. The MPEG data area **5** is for recording MPEG (Moving Picture Experts Group) video and audio data.

The lead in area **1** has records of the number of songs, the period of time of each song and total period of time, and the lead in part and index part of each song. The lead out area **6** has record of the end of the program.

As shown in FIG. 2, a conventional karaoke CD system includes a first demodulation part **11** for demodulating EFM

(Eight to Fourteen Modulation) signals. A second modulation part **12** for demodulating signals recorded in interleaved sectors for a CD-ROM upon receiving signals from the first demodulation part **11**. A CPU **16** for demultiplexing system flow on receiving signals from the second demodulation part **12**. A first, and a second memory **17** and **18** for storing video and audio data received from the CPU **16** respectively. A MPEG video decoder **19** and a MPEG audio decoder **20** for decoding MPEG video signals and MPEG audio signals of the video and audio data stored in the first and the second memories **17** and **18** respectively. A STC (System Time Clock) **23** for clocking the system time under the control of the CPU **16**. A video and audio signal processing part **13** having a video presentation part **21** for comparing the STC of the STC part **23** to PTS (Presentation Time Stamps) in the bit stream received from the MPEG video decoder **19**. An audio presentation part **22** for comparing the STC of the STC part **23** to PTS (Presentation Time Stamps) in the bit stream received from the MPEG audio decoder **20**. A control part **10** for controlling both the first and the second demodulation parts **11** and **12** and for controlling the video and audio signal processing part **13** according to a key signal applied thereto. A Digital-to-Analog (D/A) converter **14** for converting the video signal received from the video presentation part **21** and outputting the converted signal to a monitor under the control of the control part **10**, and a D/A converter **15** for converting the audio signal received from the audio presentation part **22** and applying the converted signal to a speaker under the control of the control part **10**.

The interleaving of a conventional MPEG video and MPEG audio sectors is as shown in FIG. 3.

Since the quantity of MPEG video data is much greater than the quantity of MPEG audio data, an MPEG audio sector A is periodically positioned in between MPEG video sectors V. For example, one MPEG audio sector A may be present in every four MPEG video sectors V.

Shown in Table 1 is the audio channel encoding, which is to be explained in detail hereinafter.

TABLE 1

	Stereo Music Program	Special Music Program
CH 0	Left	mono without vocal
CH 1	Right	mono with vocal

The karaoke CD format includes a stereo music program having a left audio channel for encoding on MPEG audio channel 0 and a right audio channel for encoding on MPEG audio channel 1, and a special music program having one channel for encoding music accompaniment sound without vocal and the other channel for encoding music accompaniment sound with vocal.

Operation of the conventional karaoke CD system is to be explained hereinafter.

The karaoke data is read in through a pick-up, having EFM demodulated and CD errors corrected through the first demodulation part **11** which is an EFM/CIRC (Cross Interleave Read Solomon Code) decoder. The data interleaved at encoding is demodulated through the second demodulation part **12** which is a CD-ROM decoder. The data is divided into audio data and video data through the video and audio signal processing part **13** under the demultiplexing operation of the CPU **16** and applied to the MPEG video decoder **19** and MPEG audio decoder **20** through the first and the second memories **17** and **18** respectively.

The MPEG video decoder **19** and the MPEG audio decoder **20**, restore the compressed data, and compare the



STC from the STC part 23 to the PTS in the restored data to control the presentation time of the data.

However, such a conventional karaoke-CD system has the disadvantage of being unable to obtain stereophony sound in the case where music without vocal and music with vocal of a specific music program is carried on the first channel (Left) and the second channel (Right).

#### SUMMARY OF THE INVENTION

The object of this invention is to provide a user with selective reproduction of playing sound, i.e., voice (song of a singer) and accompaniment sound.

These and other objects and features of this invention can be achieved by providing a karaoke CD format which is divided into video sectors and audio sectors, and the audio sectors are further divided into a part for recording accompaniment sound and vocal sound at the same time and another part for recording accompaniment sound only. The audio sectors are further provided to record coding information for distinguishing between the audio data stored in the two parts. There is provided a device for controlling audio signals using a karaoke CD format including a control means for controlling the entire system according to signals applied from outside the system. There is provided a demodulation means for demodulating a signal read in from a CD which distinguishes between a part containing recorded accompaniment sound and vocal sound at the same time, and another part containing recorded accompaniment sound only under the control of the control means. There is provided an audio signal processing means for processing audio signals using signals received from the demodulation means and transmitting data corresponding to a first data signal corresponding to the accompaniment sound and playing sound and a second data signal corresponding to the playing sound only, a switching means for selecting one of the first or second data signals and outputting the selected data under the control of the control means, and a D/A converting means for converting the data received from the switching means and applying the converted data to a speaker under the control of the control means.

*According to an embodiment, the present invention provides an apparatus for processing a digital data, comprising: a demultiplexing circuit to receive data units of digital audio data and separate a system time information from at least one data unit of digital audio data, the digital audio data including at least two of: 1) a first type of digital audio data, and 2) a second type of digital audio data which includes at least one content different from the first type of digital audio data; each type of digital audio data being specified by attribute information, which has: a) first information for indicating a coding mode of the digital audio data, and b) second information, different than the first information, for identifying one of the first type of digital audio data and the second type of digital audio data; a data storage to temporarily store the first and second types of digital audio data; an audio data processor to process at least one of first and second types of digital audio data identified by the second information, the audio data processor including a decoder to decode the compressed digital audio data; a system time clock circuit to receive the system time information and generate a system time clock based on the system time information; and a control circuit, coupled to the audio data processor, to control a presentation timing of the first or second types of digital audio data in response to the system time clock from the system time clock circuit.*

*According to an embodiment, the present invention provides a method for processing a digital data, comprising:*

*receiving data units of digital audio data, the data units corresponding to at least two of: 1) a first type of digital audio data, and 2) a second type of digital audio data which includes at least one content different from the first type of digital audio data; each type of digital audio data being specified by attribute information, which has: a) first information for indicating a coding mode of the digital audio data, and b) second information, different than the first information, for identifying one of the first type of digital audio data and the second type of digital audio data; separating a system time information from at least one of data units of first or second types of digital audio data and separating the first type of digital audio data from the second type of digital audio data using the second information and storing the first and second types of digital audio data; processing at least one of the first and second types of digital audio data to output the first or second types of digital audio data in response to a designation signal, the designation signal for selecting the first or second types of digital audio data; and controlling a presentation timing of the first or second types of digital audio data in response to the system time information, wherein the first or second types of digital audio data is identified by the second information, in response to the designation signal.*

#### BRIEF DESCRIPTION OF THE DRAWINGS

FIG. 1 is a layout of a karaoke CD data.

FIG. 2 is a block diagram of a conventional karaoke CD system.

FIG. 3 is a diagram showing a conventional interleaving of MPEG audio/video sectors.

FIG. 4 is a diagram of a subheader in accordance with this invention.

FIG. 5 is a diagram showing an interleaving of MPEG audio/video sectors in accordance with this invention.

FIG. 6 is a schematic block diagram of a device for controlling karaoke CD audio signal in accordance with this invention.

FIG. 7 is an operational flow diagram of a device for controlling karaoke CD audio signals.

FIG. 8 is an MPEG audio format in accordance with this invention.

#### DETAILED DESCRIPTION OF THE INVENTION

This invention is to be explained in detail hereinafter, referring to the attached drawings.

A subheader in accordance with this invention is shown in FIG. 4. To distinguish audio data Aa, containing both vocal sound and accompaniment sound, and audio data Ab, containing only accompaniment data, there is provided subheader information in an audio packet, the subheader includes coding information encoded therein with a distinguishing value applied thereto.

The coding information is, included into a subheader of each sector at data encoding as a distinguishing value, standardized as \$0 F in case of MPEG video, and \$7 F in case of MPEG audio, wherein new coding information (for example 3F) is added to use for a distinguishing signal for the audio data sector Aa, having both vocal sound and accompaniment sound, and the audio data sector Ab having only accompaniment sound.

Interleaving of MPEG video and MPEG audio sectors is as shown in FIG. 5. The interleaving of video data and audio data sectors, in general has 4 to 5 consecutive video sectors V with one audio sector A added thereto, where coding



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information is used to distinguish a sector containing the encoded Aa and Ab.

Since the increase of buffer size due to the newly added audio sectors Aa and Ab is infinitesimal compared to the image data, it can be ignored.

That is, the karaoke CD format in accordance with this invention includes a CD format divided into video sectors and audio sectors, and the audio sectors are further divided into a part Aa for recording both accompaniment sound and vocal sound at the same time and a part Ab for recording accompaniment sound only, and the audio sectors are further provided to record coding information for distinguishing between audio data stored in the two parts.

As shown in FIG. 6, a device for controlling a karaoke CD audio signal in accordance with this invention in a state having the audio data interleaved as explained above, includes a control part 30, a demodulation part 40, an audio signal processing part 70, a switch 50, and a D/A converting part 60.

The control part 30 controls the entire system according to the signal applied through a key matrix 80.

The demodulation part 40 includes a first demodulator 41 for carrying out EFM demodulation from the CD format provided to distinguish a part Aa for recording accompaniment sound and vocal sound at the same time, and a part Ab for recording accompaniment sound only, where the part is under the control of the control part 30, and a second demodulator 42 for demodulating signals recorded in interleaved sectors for a CD-ROM from the signal received from the first demodulator 41 and applying the second demodulated signal to the audio processing part 70 under the control of the control part 30, thereby demodulating the signal read in from the CD format provided to distinguish a part Aa, for recording accompaniment sound and vocal sound at the same time, and a part Ab, for recording accompaniment sound only, under the control of the control part 30.

The audio signal processing part 70 includes a CPU 71 for separating audio data and time data from the signal received from the second demodulator 42 under the control of the control part 30, a STC (System Time Clock) part 77 for generating a synchronizing signal using time information received from the CPU 71, a memory 72 for storing the audio data received from the CPU 71, an MPEG audio decoder 73 for decoding the MPEG audio data stored in the memory 72, a demultiplexer 74 for separating data received from the MPEG audio decoder 73 into data corresponding to accompaniment sound and vocal sound and data corresponding to accompaniment sound and outputting the separated data under the control of the CPU 71, a first audio presentation part 75 for comparing the PTS (Presentation Time Stamp) in the data corresponding to the accompaniment sound received from the demultiplexer 74 to the signal received from the STC 77 to control presentation time of the data, and a second audio presentation part 76 for comparing the PTS (Presentation Time Stamp) in the data corresponding to accompaniment sound and vocal sound received from the demultiplexer 74 to the signal received from the STC 77 to control presentation time of the data, thereby processing the audio signal using signals received from the demodulation part 40 to separate the data into data corresponding to a first data containing accompaniment sound only, and outputting the separate data.

The switch 50 selects one of the data corresponding to the accompaniment sound and vocal sound, or to only the accompaniment sound received from the first and the second audio presentation parts 75 and 76 of the audio signal processing part 70 and transmits the selected data under the control of the control part 30.

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The D/A converter 60 converts the data received from the switch 50 under the control of the control part 30.

The system and operation for processing a video signal is the same as for a conventional karaoke CD system.

Operation of the device for controlling a karaoke CD audio signal in accordance with this invention as described above is to be explained hereinafter.

The encoded karaoke-CD data having coding information applied thereto is added with a distinguishing value and has EFM signals demodulated through the first demodulation part 41 which is a EFM/CIRC (Cross Interleave Read Solomon Code) and errors of the CD are corrected. The interleaved data at encoding is demodulated through the second demodulation part 42 which is a CD-ROM decoder, and is separated into audio, video, and time data in the CPU 71.

In the instant, of audio data, the demultiplexing signal coming in at the reading in of the coding information is applied to the demultiplexer 74 for operation of the demultiplexer 74.

Of the data demultiplexed in the CPU 71, the MPEG audio data is stored in the memory 72 temporarily, has the compression restored through the MPEG audio decoder 73, and is transmitted. The demodulated audio data received from the MPEG audio decoder 73 is separated into audio data carrying accompaniment sound only and audio data carrying accompaniment sound and vocal sound through the demultiplexer 74, and applied to the first audio data presentation part 75 and the second audio presentation part 76 respectively under the control of the CPU 71.

The audio demultiplexing signal received from the CPU 71 controls the first audio presentation part 75 to compare the received audio data Ab, carrying accompaniment sound only out of the audio signal applied thereto, to a PTS to control the presentation time and controls the second audio presentation part 76 to compare the received audio data Aa, carrying voice, i.e., vocal sound and accompaniment sound out of the audio signal applied thereto to a PTS to control the presentation time. The audio data received from the first and the second audio presentation parts 75 and 76 is applied to the D/A converter 60 through a switching operation of the switch 50 under the control of the control part 30, where it is converted in the D/A converter 60 and output to a speaker.

In this instant, the switch 50 is operated under the control of the control part 30 which is operated according to the key signal applied through the key matrix 80.

Operation of a karaoke CD audio system is to be explained hereinafter, referring to FIG. 7.

First, a subheader and packet header are read 100, and demultiplexing is carried out to separate into an audio demultiplexing signal and a MPEG audio signal 101. The separated MPEG audio signal is decoded to separate into audio signals containing accompaniment sound only, and audio signals containing both vocal sound and accompaniment sound using the audio demultiplexing signal 102, and the PTS of each packet is read to synchronize and transmit signals 103.

The MPEG audio format, as shown in FIG. 8, generates separated MPEG audio signals Aa and Ab, and audio demultiplexing signals by extracting coding information from the subheaders of each sector, and by reading the packet header and comparing it to the STC, thereby the presentation time of the data can be controlled.

As has been explained, this invention for a karaoke CD system increases the enjoyment of karaoke by reproducing audio data containing a player's voice, i.e., vocal sound and



accompaniment sound, and audio data containing accompaniment sound selectively, which permits stereophony sound corresponding to the CD sound.

Although the invention has been described in conjunction with specific embodiments, it is evident that many alternatives and variations will be apparent to those skilled in the art in light of the foregoing description. Accordingly, the invention is intended to embrace all of the alternatives and variations that fall within the spirit and scope of the appended claims.

What is claimed is:

**[1.** A device for reproducing a digital signal recorded on a medium, the digital signal including a video signal and an audio signal, the audio signal being composed of data units, the data units corresponding to at least one of:

- 1) a first type of digital audio data, and
- 2) a second type of digital audio data which includes at least one content different from the first type of digital audio data;

each data unit including attribute information, which has:

- a) first information indicating a coding mode of the digital audio data, and
- b) second information, different than said first information, identifying the data unit as one of the first type of digital audio data or the second type of digital audio data;

the data units of audio signal being interleaved between data units of video signal, comprising:

a demodulator for demodulating the digital signal to restore an original signal;

a signal processor for receiving the audio signal, demodulated by the demodulator, extracting the second information, separating the data units corresponding to a first type of audio signal from data units corresponding to a second type of audio signal based on the extracted second information, said signal processor including a system time clock, a memory for storing the data units corresponding to the first type and the second type of audio signal, and first or second audio presentation parts receiving the system time clock to control a presentation timing of the first type or the second type of audio signal; and

a controller, coupled to the signal processor, controlling the signal processor to output the data units corresponding to the first type or the second type of the audio signal designated by a user input, wherein the first or second type of audio signal, designated by the user input, is identified using the second information.]

**[2.** A device as claimed in claim 1, wherein the first type of audio signal corresponds to accompaniment sound.]

**[3.** A device as claimed in claim 1, wherein the audio signal is encoded by an MPEG coding mode, wherein the signal processor further comprises:

an MPEG audio decoder for decoding the audio signal.]

**[4.** A method for reproducing a digital signal recorded on a medium, said digital signal including a video signal and an audio signal, the audio signal being composed of data units, the data units corresponding to at least one of:

- 1) a first type of digital audio data, and
- 2) a second type of digital audio data which includes at least one content different from the first type of digital audio data;

each data unit including attribute information, which has:

a) first information indicating a coding mode of the digital audio data, and

b) second information, different than said first information, identifying the data unit as one of the first type of digital audio data or the second type of digital audio data;

the data units of audio signal being interleaved between data units of video signal, the method comprising the steps of:

demodulating the digital signal to restore an original signal;

receiving the demodulated audio signal;

extracting the second information;

separating the data units corresponding to a first type of audio signal from data units corresponding to a second type of audio signal based on the extracted second information, by storing the first and second types of audio signals in a memory, receiving the data units corresponding to the first and second types of audio signals in first and second audio presentation parts along with a system time clock to control a presentation timing of the first or second type of audio signal; and

outputting one of the first or second types of audio signals in response to a user input designating one of the first or second types of audio signals, wherein the first or second type of audio signal, designated by the user input, is identified using the second information.]

**[5.** The method of claim 4, wherein the step of separating includes separating the first type of the audio signal representing accompaniment sound.]

**[6.** A device for processing a digital signal, comprising:

an audio signal processor receiving data units of digital audio data, the data units corresponding to at least one of:

- 1) a first type of digital audio data, and
- 2) a second type of digital audio data which includes at least one content different from the first type of digital audio data;

each data unit including attribute information, which has:

- a) coding information indicating a coding mode of the digital audio data, and
- b) identification information, different than said coding information, identifying the data unit as one of the first type of digital audio data or the second type of digital audio data;

the audio signal processor extracting the identification information, and separating the data units of the first type from the data units of the second type using the identification information, said signal processor including a system time clock, a memory for storing the data units of the first type and the second type, and first and second audio presentation parts receiving the system time clock to control a presentation timing of the data units of the first type or the second type; and a control circuit controlling the audio signal processor to output an audio signal corresponding to the data units of one of the first or second types, based on the identification information.]

**[7.** The device of claim 6, wherein the audio data of the first type includes accompaniment sound.]

**[8.** The device of claim 6, wherein the data units of digital audio data are interleaved with digital video data, and wherein the audio data of the first type includes accompaniment sound and vocals, which are associated with the digital video data.]



**[9.** The device of claim 6, wherein the audio signal processor compares the system time clock to timing information in the digital audio data, and outputs audio data with a timing based on the comparison.]

**[10.** The device of claim 6, wherein the audio signal processor MPEG decodes the audio data.]

**[11.** The device of claim 6, further comprising:  
a demodulator demodulating the digital audio data prior to receipt by the audio signal processor.]

**[12.** A method for processing a digital signal, comprising:  
receiving data units of digital audio data, the data units corresponding to at least one of:

- 1) a first type of digital audio data, and
  - 2) a second type of digital audio data which includes at least one content different from the first type of digital audio data;
- each data unit including attribute information, which has:

- a) coding information indicating a coding mode of the digital audio data, and
- b) identification information, different than said coding information, identifying the data unit as one of the first type of digital audio data or the second type of digital audio data;

extracting the identification information;  
separating the data units of the first type from the data units of the second type using the identification information by storing the data units of the first and second types in a memory, receiving the data units of the first and second types in first or second audio presentation parts along with a system time clock to control a presentation timing of data units of the first or second type; and

outputting audio data corresponding to the data units of the first or second type in response to the identification information.]

**[13.** The method of claim 12, wherein the audio data of the first type includes accompaniment sound.]

**[14.** The method of claim 12, wherein the data units of digital audio data are interleaved with digital video data, and wherein the audio data of the first type includes accompaniment sound and vocals, which are associated with the digital video data.]

**[15.** The method of claim 12, further comprising:  
comparing the system time clock to timing information in the digital audio data; and  
outputting the audio data with a timing based on the comparison.]

**[16.** The device of claim 12, wherein the separating step includes MPEG decoding the audio data.]

**[17.** The device of claim 1, wherein the signal processor includes a switch that selectively outputs the data units of the first type or the second type.]

*18. An apparatus for processing a digital data, comprising:*

*a demultiplexing circuit to receive data units of digital audio data and separate a system time information from at least one data unit of digital audio data, the digital audio data including at least two of:*

- 1) a first type of digital audio data, and*
- 2) a second type of digital audio data which includes at least one content different from the first type of digital audio data;*

*each type of digital audio data being specified by attribute information, which has:*

- a) first information for indicating a coding mode of the digital audio data, and*

*b) second information, different than said first information, for identifying one of the first type of digital audio data and the second type of digital audio data;*

*a data storage to temporarily store the first and second types of digital audio data;*

*an audio data processor to process at least one of first and second types of digital audio data identified by the second information, the audio data processor including a decoder to decode the compressed digital audio data;*

*a system time clock circuit to receive the system time information and generate a system time clock based on the system time information; and*

*a control circuit, coupled to the audio data processor, to control a presentation timing of the first or second types of digital audio data in response to the system time clock from the system time clock circuit.*

*19. The apparatus of claim 18, wherein the demultiplexing circuit is further configured to separate the first type of digital audio data from the second type of digital audio data by using the second information.*

*20. The apparatus of claim 18, wherein the data units of digital audio data are multiplexed with data units of digital video data, wherein the demultiplexing circuit is further configured to separate the digital audio data from the digital video data, by using the first information.*

*21. The apparatus of claim 18, wherein the control circuit is configured to control the presentation timing based on a presentation time information included in the digital audio data, the presentation time information being synchronized to the system time clock from the system time clock circuit, and control the output of the first or second types of digital audio data with a timing.*

*22. The apparatus of claim 21, wherein the demultiplexing circuit is further configured to separate the presentation time information from at least one data unit of first or second types of digital audio data.*

*23. The apparatus of claim 18, wherein the control circuit is further configured to identify the coding mode of the first or second types of digital audio data, the coding mode for indicating MPEG audio data, and wherein the decoder is configured to decode the digital audio data compressed in MPEG format.*

*24. The apparatus of claim 18, further comprising:  
a demodulator to demodulate the digital audio data to an original data, prior to receipt by the demultiplexing circuit.*

*25. The apparatus of claim 18, wherein the control circuit is further configured to receive a request for designating one of first and second types of digital audio data, and*

*wherein the first or second type of audio data, designated by the request, is identified using the second information.*

*26. The apparatus of claim 18, further comprising:*

*a D/A converter coupled to the audio data processor to convert at least one of the first and second types of digital audio data to analog signal.*

*27. The apparatus of claim 18, wherein the at least one of the first and second types of digital audio data includes vocal data, and*

*wherein the audio data processor is configured to process the vocal data.*

*28. A method for processing a digital data, comprising:  
receiving data units of digital audio data, the data units corresponding to at least two of:*



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- 1) a first type of digital audio data, and
- 2) a second type of digital audio data which includes at least one content different from the first type of digital audio data;

each type of digital audio data being specified by attribute information, which has:

- a) first information for indicating a coding mode of the digital audio data, and
- b) second information, different than said first information, for identifying one of the first type of digital audio data and the second type of digital audio data;

separating a system time information from at least one of data units of first or second types of digital audio data and separating the first type of digital audio data from the second type of digital audio data using the second information and storing the first and second types of digital audio data;

processing at least one of the first and second types of digital audio data to output the first or second types of digital audio data in response to a designation signal, the designation signal for selecting the first or second types of digital audio data; and

controlling a presentation timing of the first or second types of digital audio data in response to the system time information,

wherein the first or second types of digital audio data is identified by the second information, in response to the designation signal.

29. The method of claim 28, wherein the data units of digital audio data are multiplexed with data units of digital video data, further comprising:

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separating the digital audio data from the digital video data, by using the first information.

30. The method of claim 28, wherein the controlling step controls the presentation timing according to a presentation time information included in the digital audio data, the presentation time information being synchronized to a system time clock derived by the system time information, and outputs the first or second types of digital audio data with a timing.

31. The method of claim 28, further comprising:

identifying the coding mode of the first or second types of digital audio data based on the first information, the coding mode for indicating MPEG audio data, and wherein the processing step includes decoding the digital audio data encoded in MPEG format.

32. The method of claim 28, further comprising:

demodulating the digital audio data to an original data, prior to receipt by the receiving step.

33. The method of claim 28, further comprising:

converting the designated first or second types of digital audio data to analog signal.

34. The method of claim 28, wherein the at least one of the first and second types of digital audio data includes vocal data, and

wherein the processing step processes the vocal data.

\* \* \* \* \*