US 20100260360A1

(19) **United States**

(12) **Patent Application Publication**

Strub

(10) **Pub. No.: US 2010/0260360 A1**

(43) **Pub. Date: Oct. 14, 2010**

(54) **SYSTEMS, METHODS, AND APPARATUS FOR CALIBRATING SPEAKERS FOR THREE-DIMENSIONAL ACOUSTICAL REPRODUCTION**

(75) Inventor: **Tyner Brentz Strub**, Bonita Springs, FL (US)

Correspondence Address:
SUTHERLAND ASBILL & BRENNAN LLP
999 PEACHTREE STREET, N.E.
ATLANTA, GA 30309 (US)

(73) Assignee: **STRUBWERKS LLC**, Bonita Springs, FL (US)

(21) Appl. No.: **12/759,351**

(22) Filed: **Apr. 13, 2010**

Related U.S. Application Data

(60) Provisional application No. 61/169,044, filed on Apr. 14, 2009.

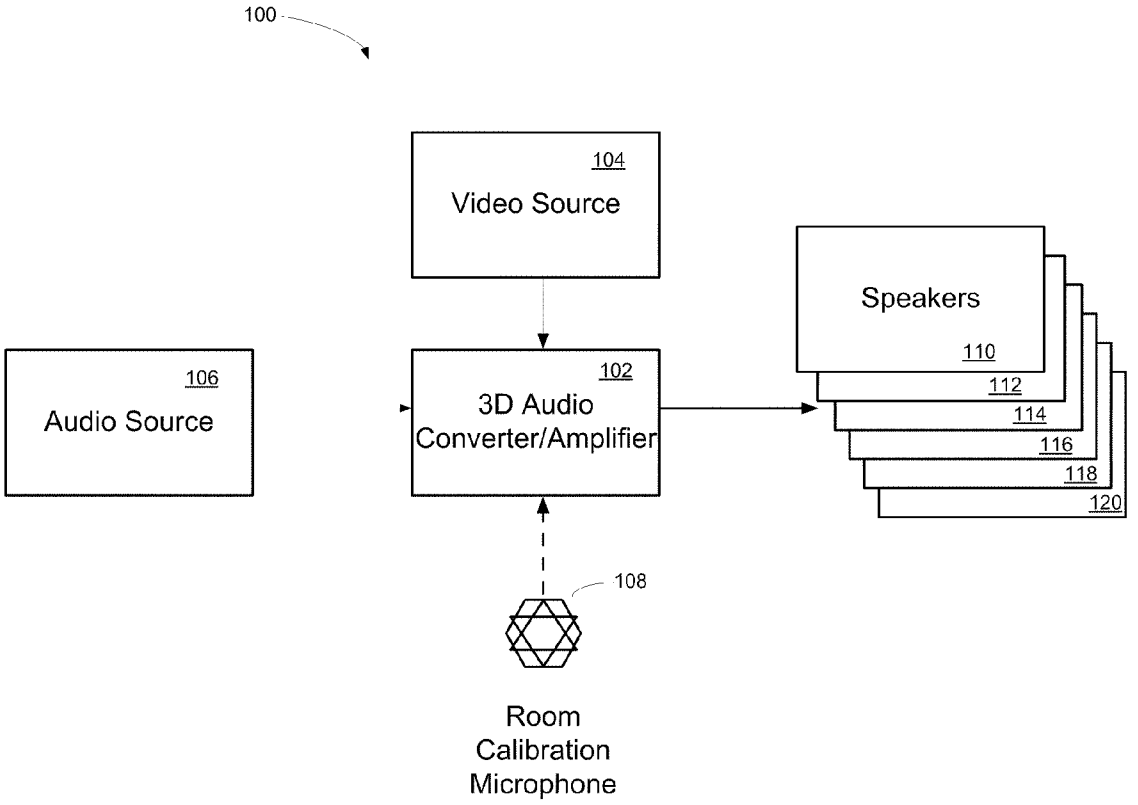
Publication Classification

(51) **Int. Cl.**
H04R 5/02 (2006.01)

(52) **U.S. Cl.** **381/303**

(57) **ABSTRACT**

Certain embodiments of the invention may include systems, methods, and apparatus for calibrating speakers for three dimensional acoustical reproduction. According to an example embodiment of the invention, a method is provided for initializing a plurality of speakers in a 3D acoustical reproduction system. The method can include positioning one or more calibration microphones near a listener position, generating calibration tones, selectively routing calibration tones to one or more of the plurality of speakers, producing audible tones from the plurality of speakers based on the generated calibration tones, sensing audible tones from the plurality of speakers with the one or more calibration microphones, determining one or more parameters associated with the plurality of speakers based on sensing the audible tones, and modifying settings of the 3D acoustical reproduction system based on the one or more determined parameters.



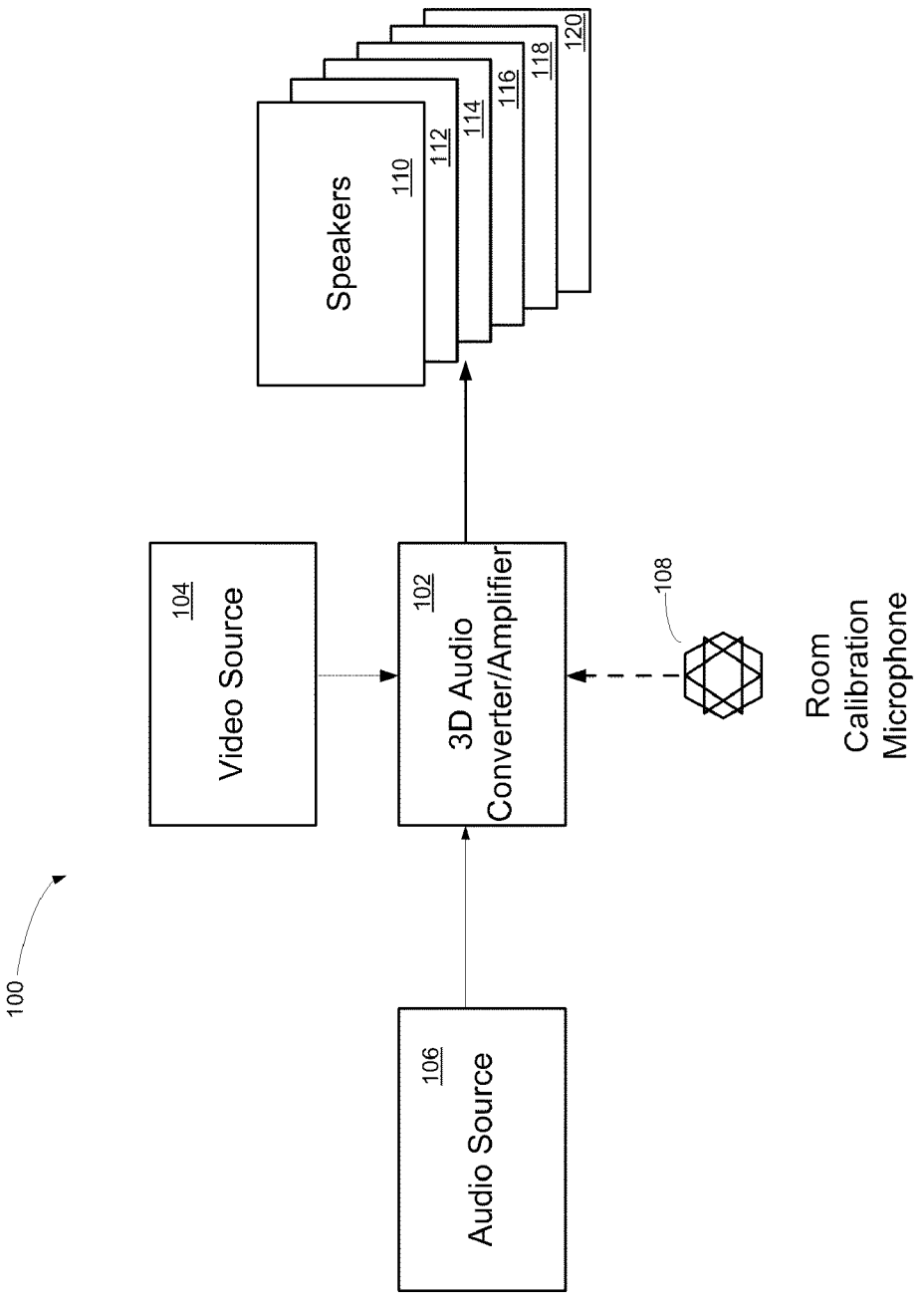

100
Audio Source
106
3D Audio Converter/Amplifier
102
Video Source
104
Speakers
110
112
114
116
118
120
108
Room Calibration Microphone

FIG. 1

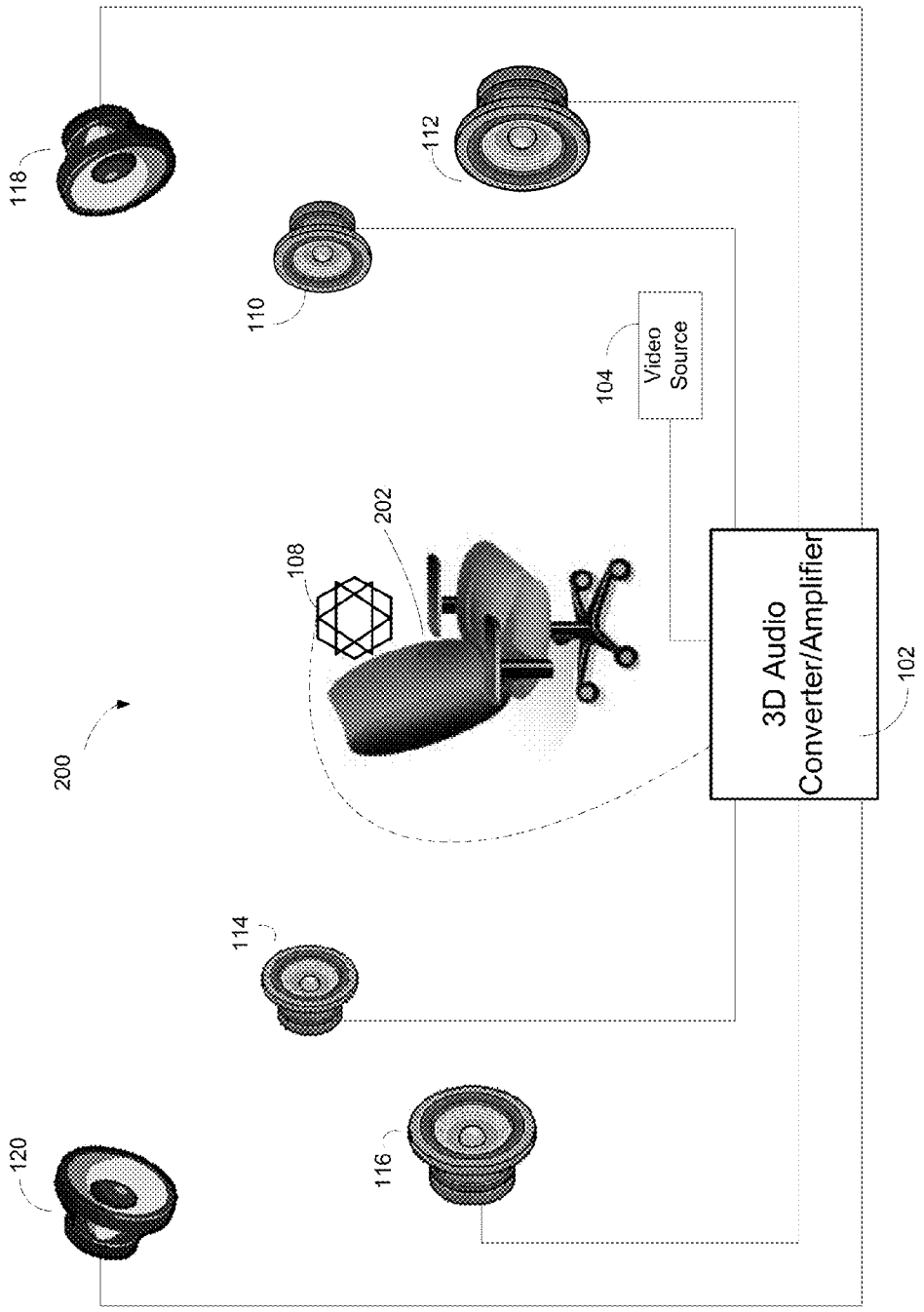

118
112
110
104
Video Source
202
108
200
3D Audio Converter/Amplifier
102
114
116
120

FIG. 2

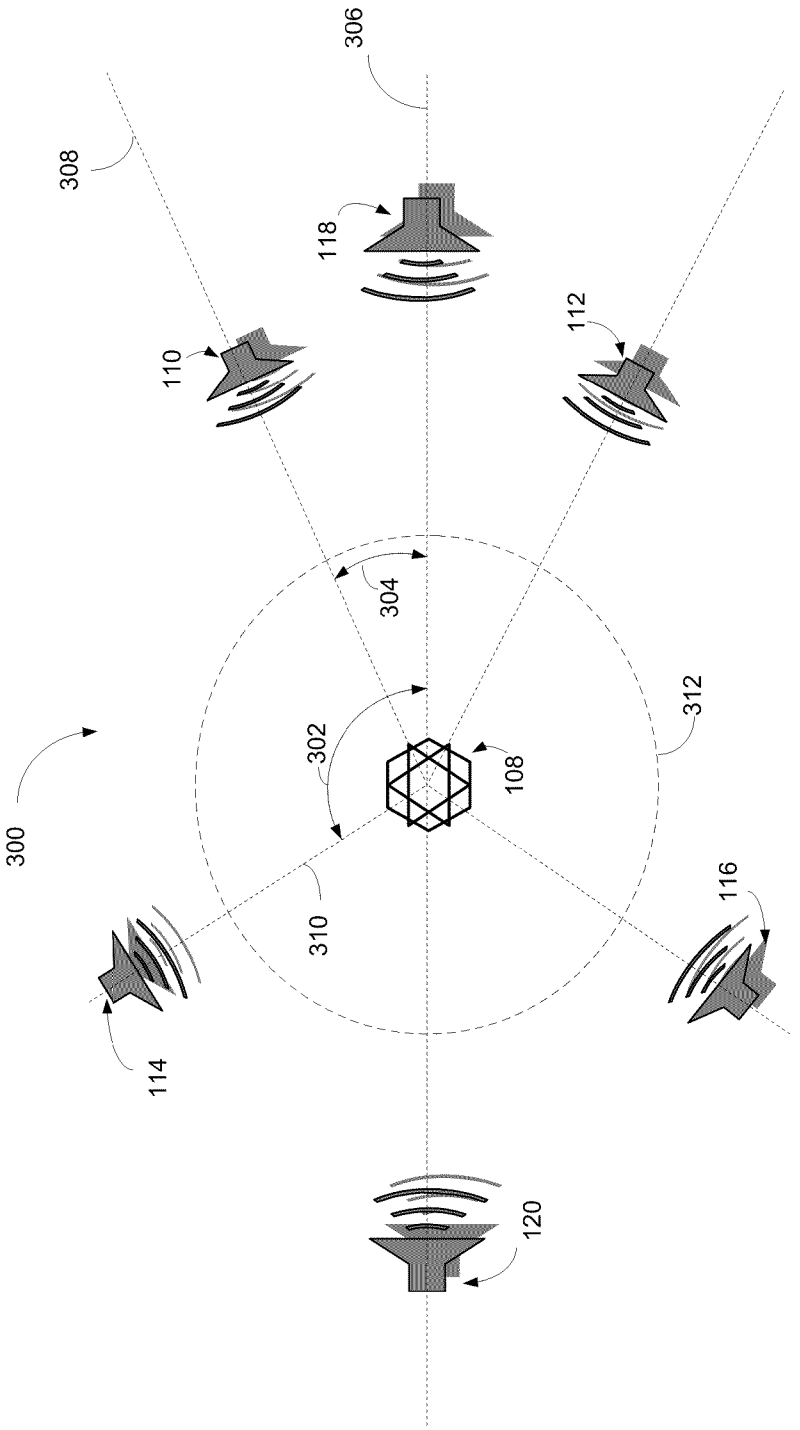


FIG. 3

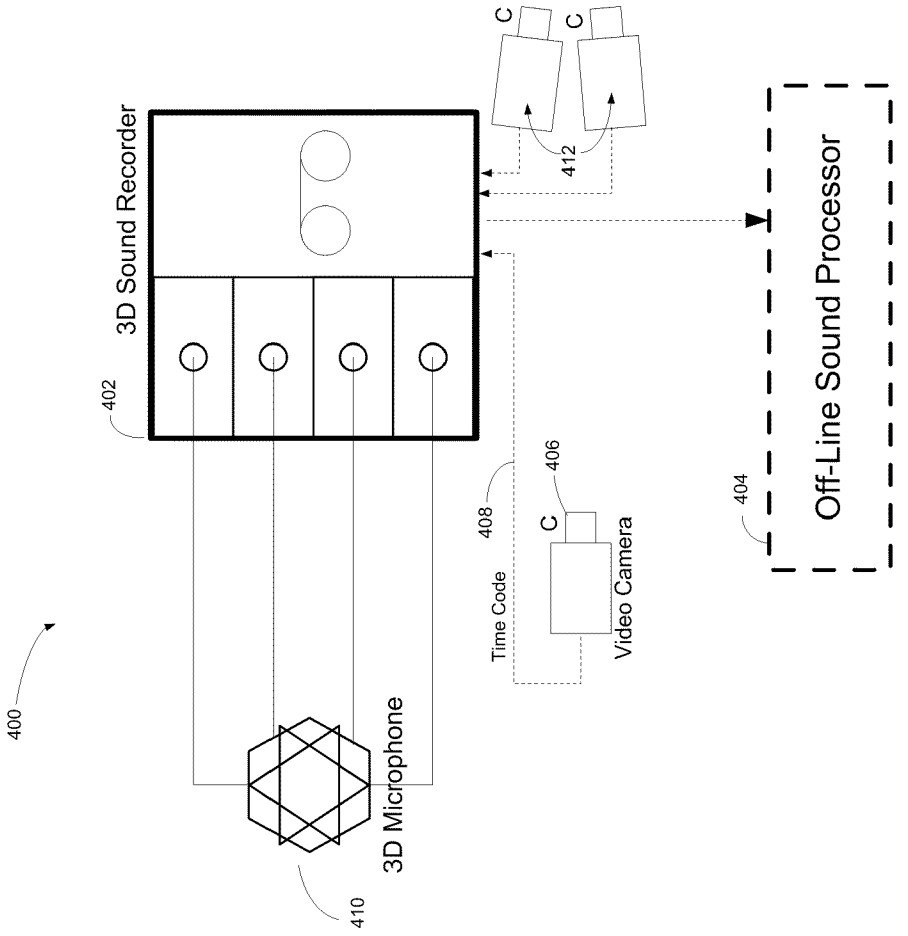

400
3D Sound Recorder
402
C
C
412
404
Off-Line Sound Processor
408
406
C
Time Code
Video Camera
3D Microphone
410

FIG. 4

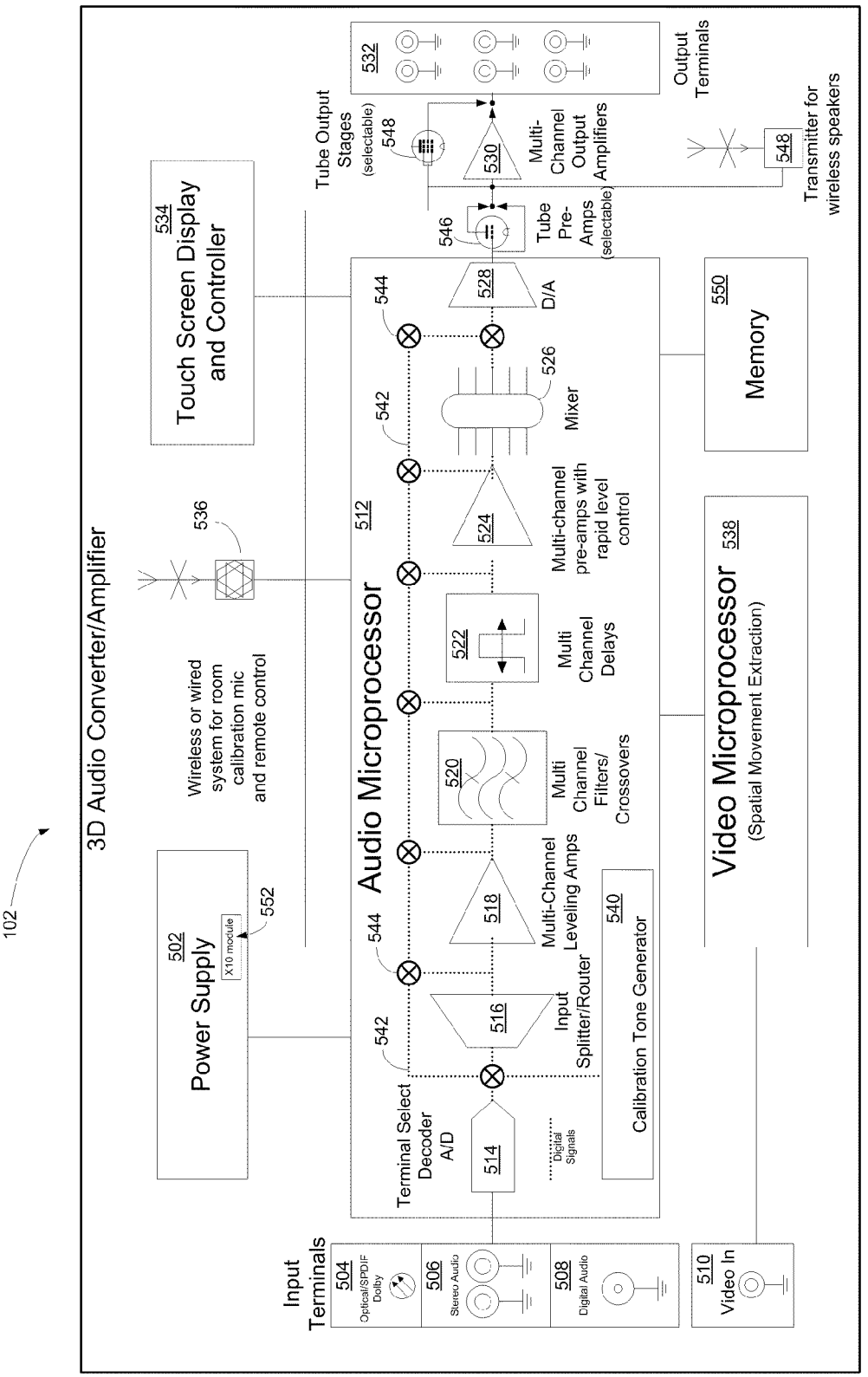

102
3D Audio Converter/Amplifier
Input Terminals
504 Optical/SPDIF Dolby
506 Stereo Audio
508 Digital Audio
510 Video In
Terminal Select Decoder A/D
514
516
Input Splitter/Router
518
Multi-Channel Leveling Amps
520
Multi Channel Filters/ Crossovers
522
Multi Channel Delays
524
Multi-channel pre-amps with rapid level control
526
Mixer
528
D/A
530
Multi-Channel Output Amplifiers
532
Output Terminals
546
Tube Pre-Amps (selectable)
548
Transmitter for wireless speakers
Tube Output Stages (selectable)
Audio Microprocessor
512
542
544
Digital Signals
Calibration Tone Generator
540
Video Microprocessor
(Spatial Movement Extraction)
538
Memory
550
Touch Screen Display and Controller
534
Wireless or wired system for room calibration mic and remote control
536
Power Supply
502
X10 module
552

FIG. 5

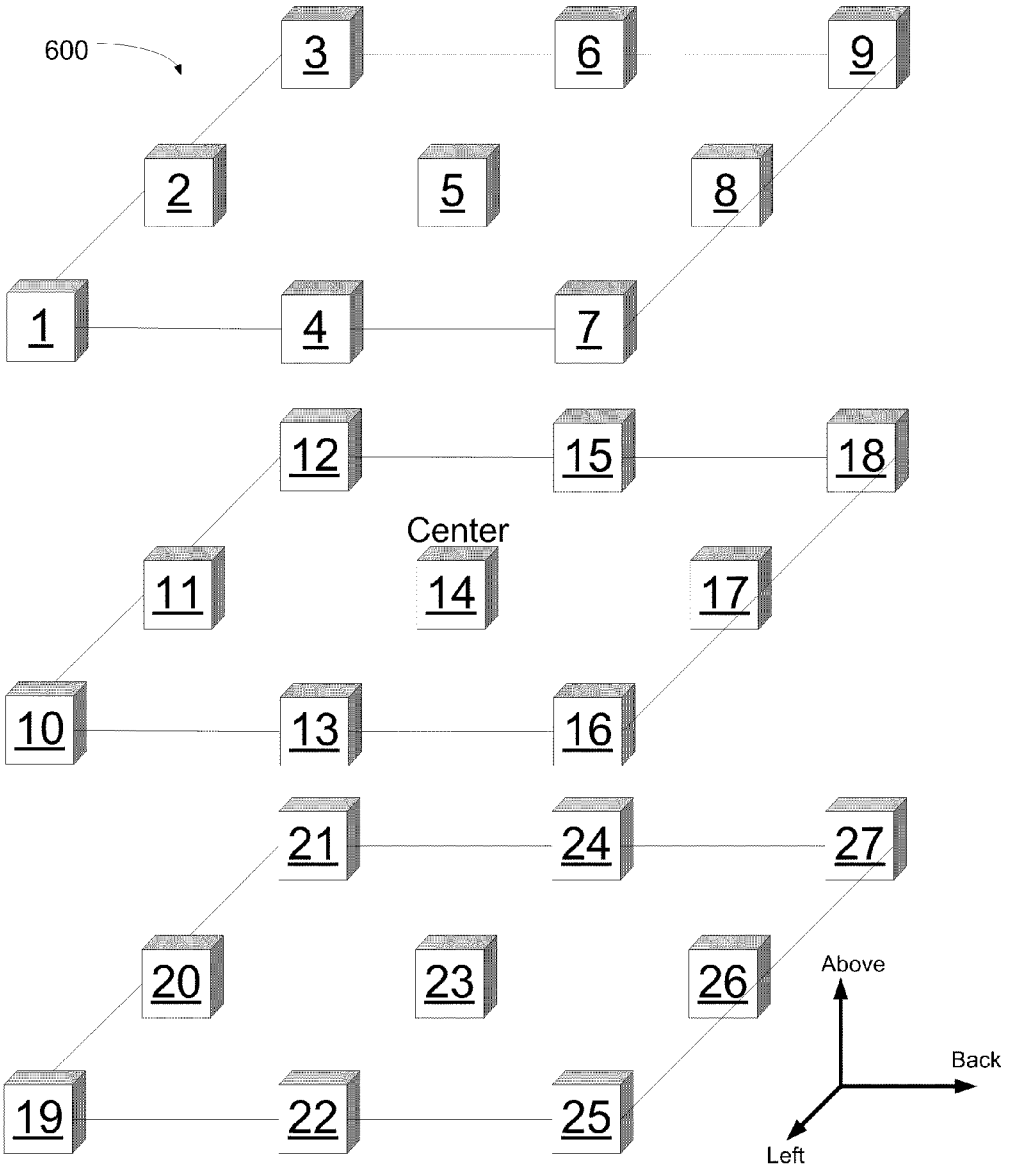

600
3
6
9
2
5
8
1
4
7
12
15
18
Center
11
14
17
10
13
16
21
24
27
20
23
26
19
22
25
Above
Back
Left

FIG. 6

spkrs/map	Left	Right	L. Surr.	R. Surr.	Top Front	Top Rear
1	+	0	0	0	+	0
2	+	+	-	-	+	0
3	0	+	-	0	+	0
4	+	-	+	-	+	+
5	0	0	0	0	+	+
6	-	+	-	+	+	+
7	0	-	+	0	0	+
8	-	-	+	+	-	+
9	-	0	0	+	0	+
10	+	0	0	-	0	-
11	+	+	-	-	0	0
12	0	+	-	0	0	-
13	+	-	+	-	0	0
14	0	0	0	0	0	0
15	-	+	-	+	0	0
16	0	-	+	0	-	0
17	-	-	+	+	0	0
18	-	0	0	+	-	0
19	+	0	0	-	-	off
20	+	+	-	-	0	-
21	0	+	-	0	-	off
22	+	-	+	-	-	-
23	0	0	0	0	-	-
24	-	+	-	+	-	-
25	0	-	+	0	off	-
26	-	-	+	+	-	0
27	-	0	0	+	off	-

FIG. 7

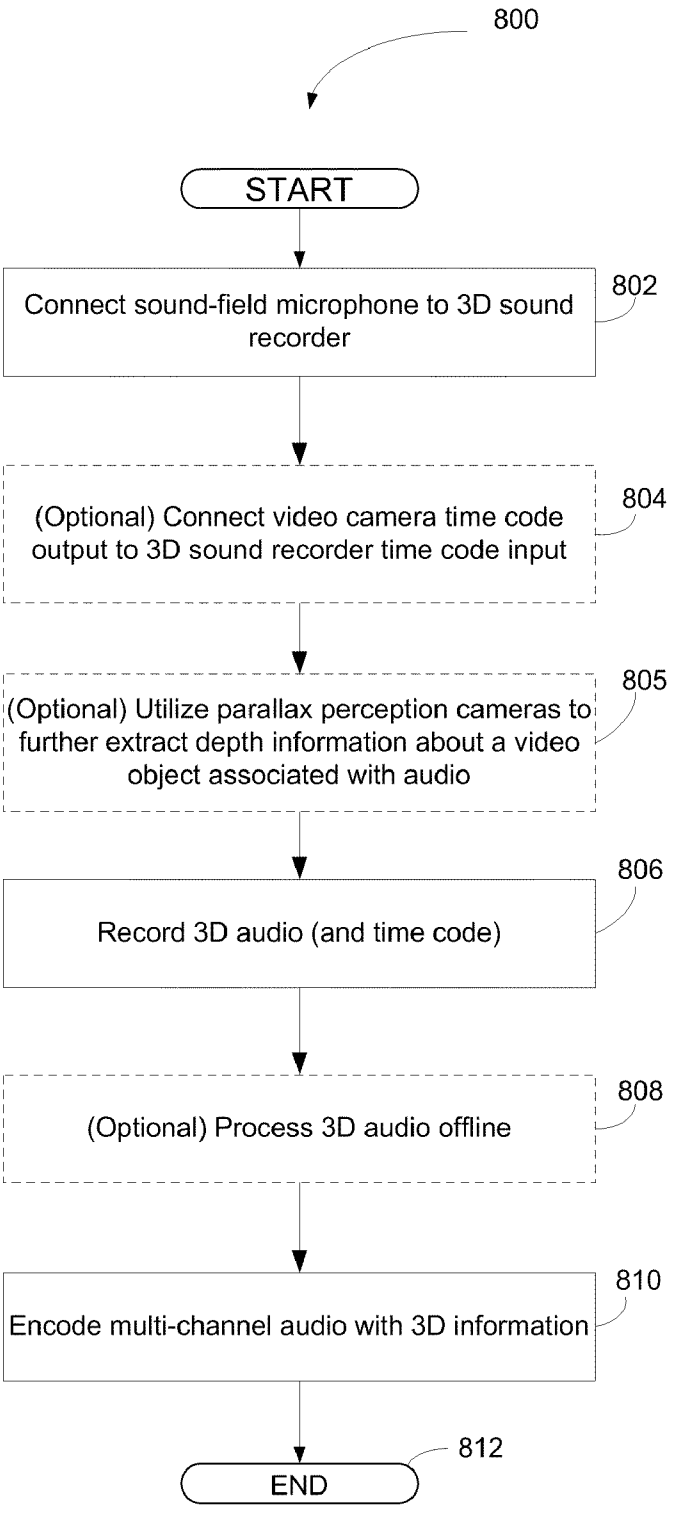

800
START
Connect sound-field microphone to 3D sound recorder
802
(Optional) Connect video camera time code output to 3D sound recorder time code input
804
(Optional) Utilize parallax perception cameras to further extract depth information about a video object associated with audio
805
Record 3D audio (and time code)
806
(Optional) Process 3D audio offline
808
Encode multi-channel audio with 3D information
810
END
812

FIG. 8

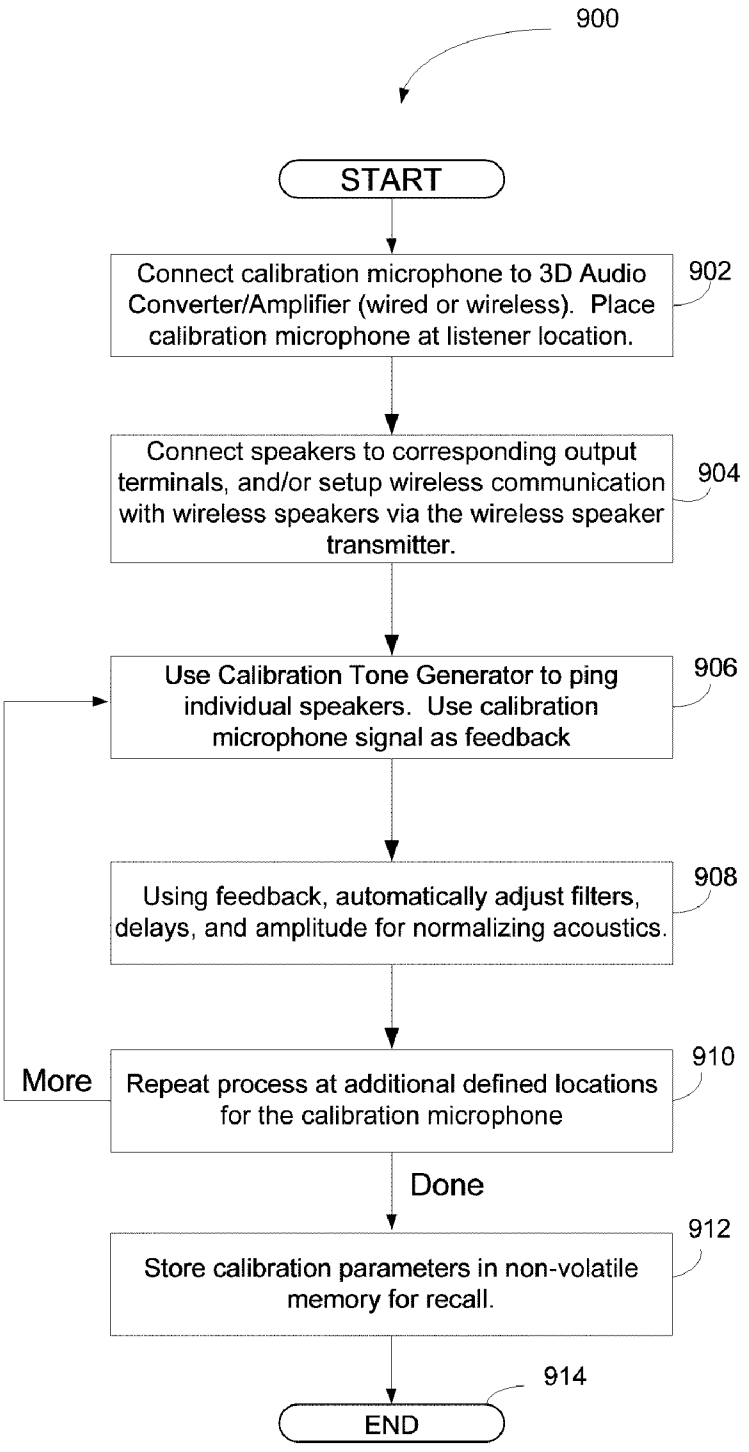

900
START
Connect calibration microphone to 3D Audio Converter/Amplifier (wired or wireless). Place calibration microphone at listener location.
902
Connect speakers to corresponding output terminals, and/or setup wireless communication with wireless speakers via the wireless speaker transmitter.
904
Use Calibration Tone Generator to ping individual speakers. Use calibration microphone signal as feedback
906
Using feedback, automatically adjust filters, delays, and amplitude for normalizing acoustics.
908
More
Repeat process at additional defined locations for the calibration microphone
910
Done
Store calibration parameters in non-volatile memory for recall.
912
914
END

FIG. 9

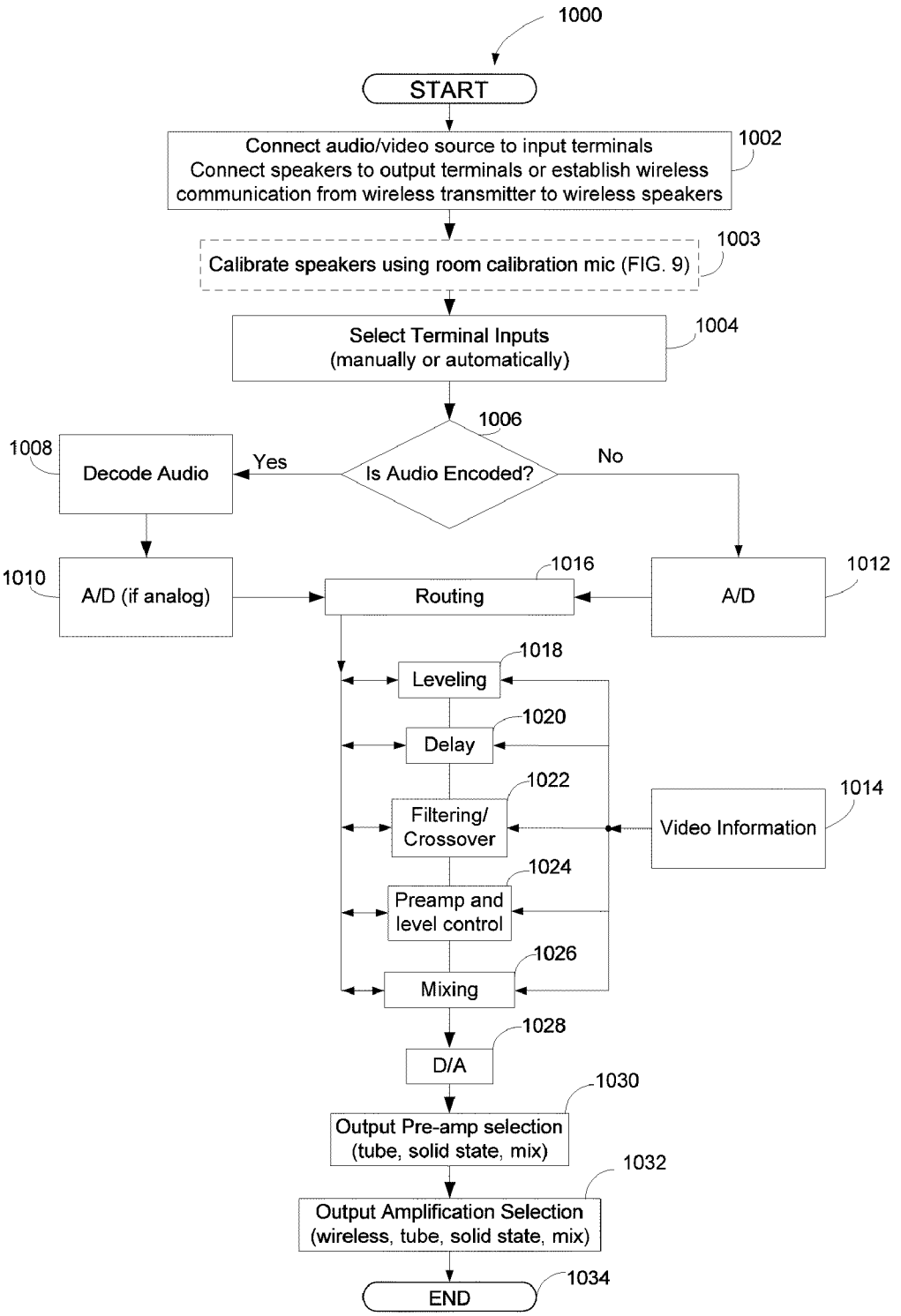

1000
START
Connect audio/video source to input terminals
Connect speakers to output terminals or establish wireless communication from wireless transmitter to wireless speakers
1002
Calibrate speakers using room calibration mic (FIG. 9)
1003
Select Terminal Inputs
(manually or automatically)
1004
1006
Is Audio Encoded?
Yes
No
1008
Decode Audio
1010
A/D (if analog)
1012
A/D
1016
Routing
1018
Leveling
1020
Delay
1022
Filtering/
Crossover
1014
Video Information
1024
Preamp and
level control
1026
Mixing
1028
D/A
1030
Output Pre-amp selection
(tube, solid state, mix)
1032
Output Amplification Selection
(wireless, tube, solid state, mix)
1034
END

FIG. 10

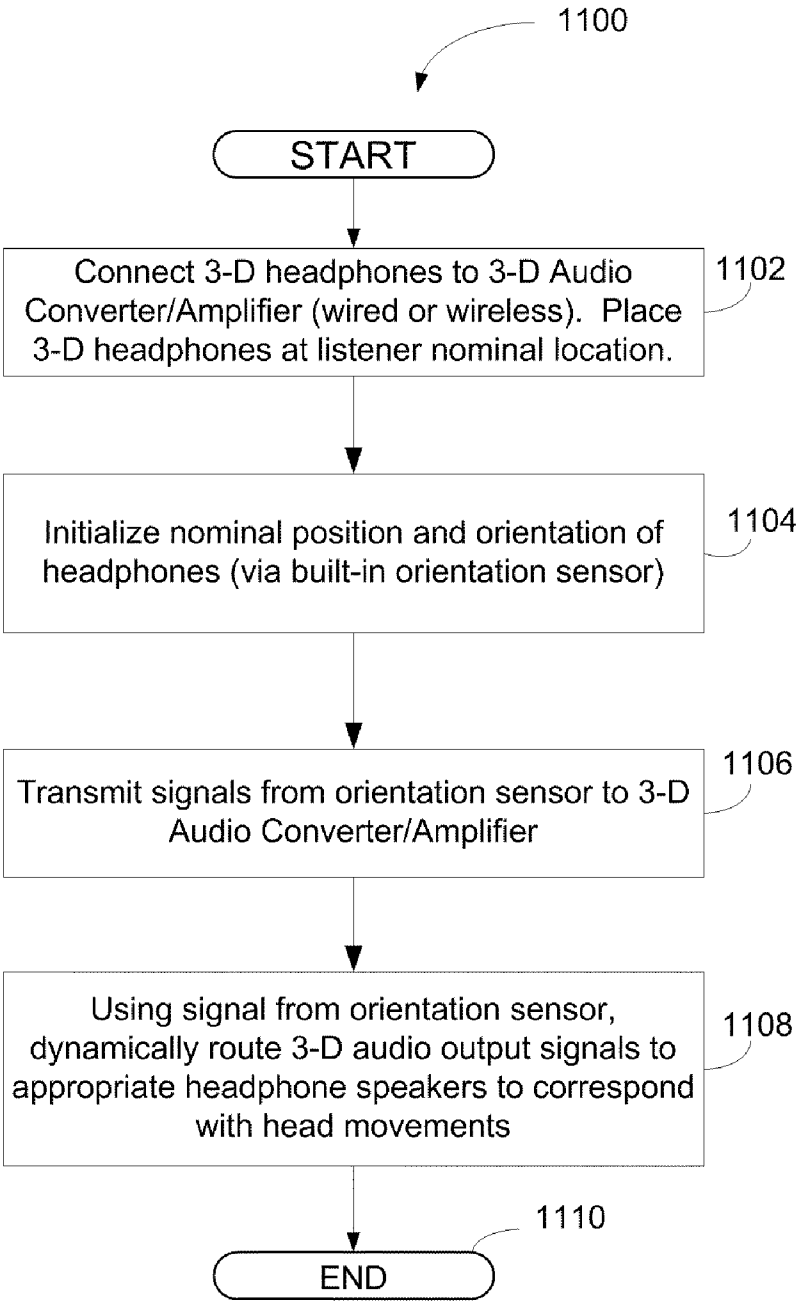

1100
START
Connect 3-D headphones to 3-D Audio Converter/Amplifier (wired or wireless). Place 3-D headphones at listener nominal location.
1102
Initialize nominal position and orientation of headphones (via built-in orientation sensor)
1104
Transmit signals from orientation sensor to 3-D Audio Converter/Amplifier
1106
Using signal from orientation sensor, dynamically route 3-D audio output signals to appropriate headphone speakers to correspond with head movements
1108
1110
END

FIG.11

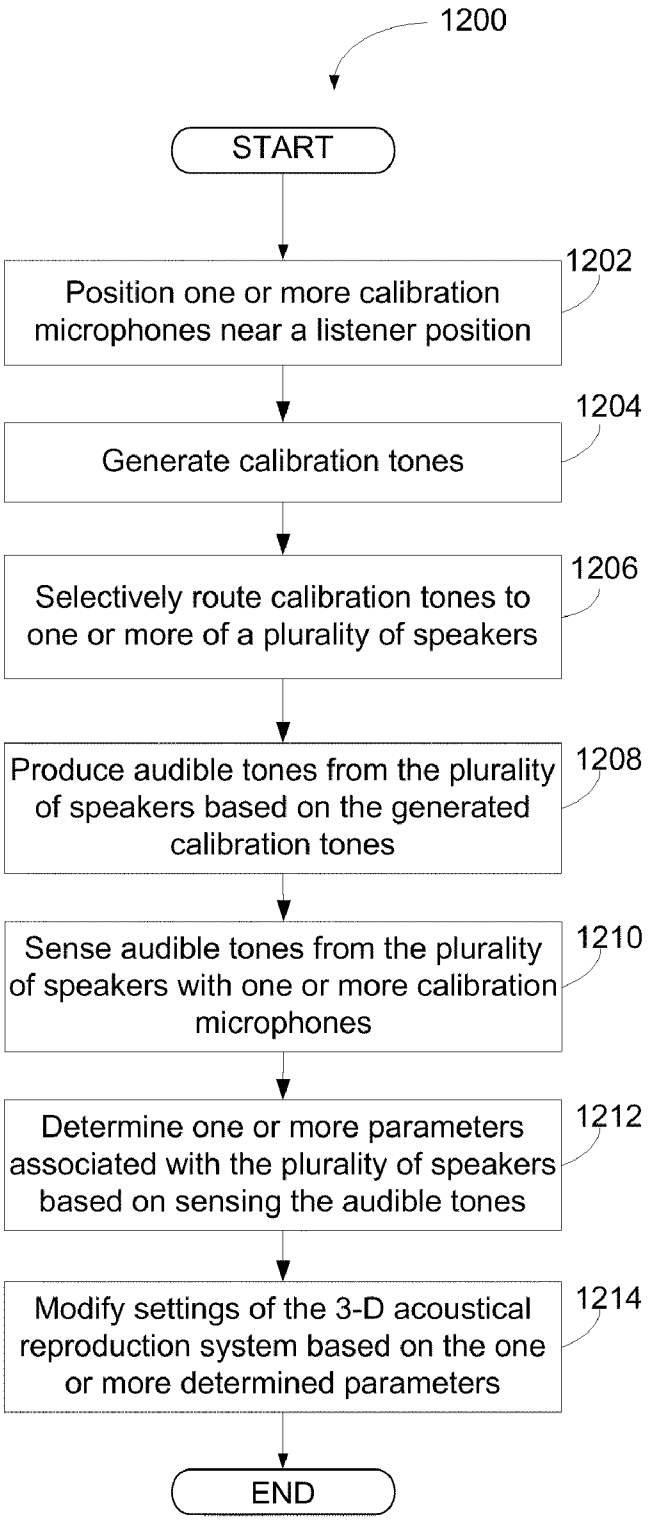

1200
START
Position one or more calibration microphones near a listener position
1202
Generate calibration tones
1204
Selectively route calibration tones to one or more of a plurality of speakers
1206
Produce audible tones from the plurality of speakers based on the generated calibration tones
1208
Sense audible tones from the plurality of speakers with one or more calibration microphones
1210
Determine one or more parameters associated with the plurality of speakers based on sensing the audible tones
1212
Modify settings of the 3-D acoustical reproduction system based on the one or more determined parameters
1214
END

FIG.12

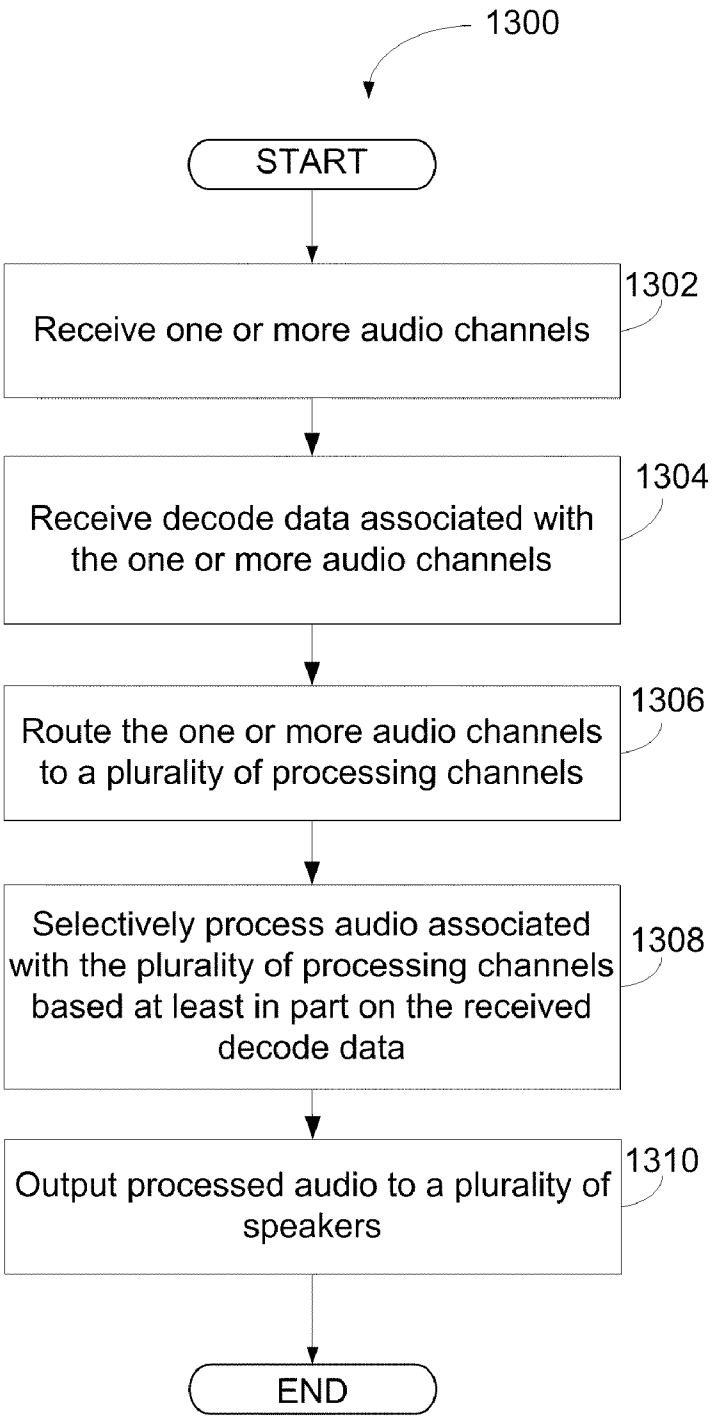

1300
START
Receive one or more audio channels
1302
Receive decode data associated with the one or more audio channels
1304
Route the one or more audio channels to a plurality of processing channels
1306
Selectively process audio associated with the plurality of processing channels based at least in part on the received decode data
1308
Output processed audio to a plurality of speakers
1310
END

FIG.13

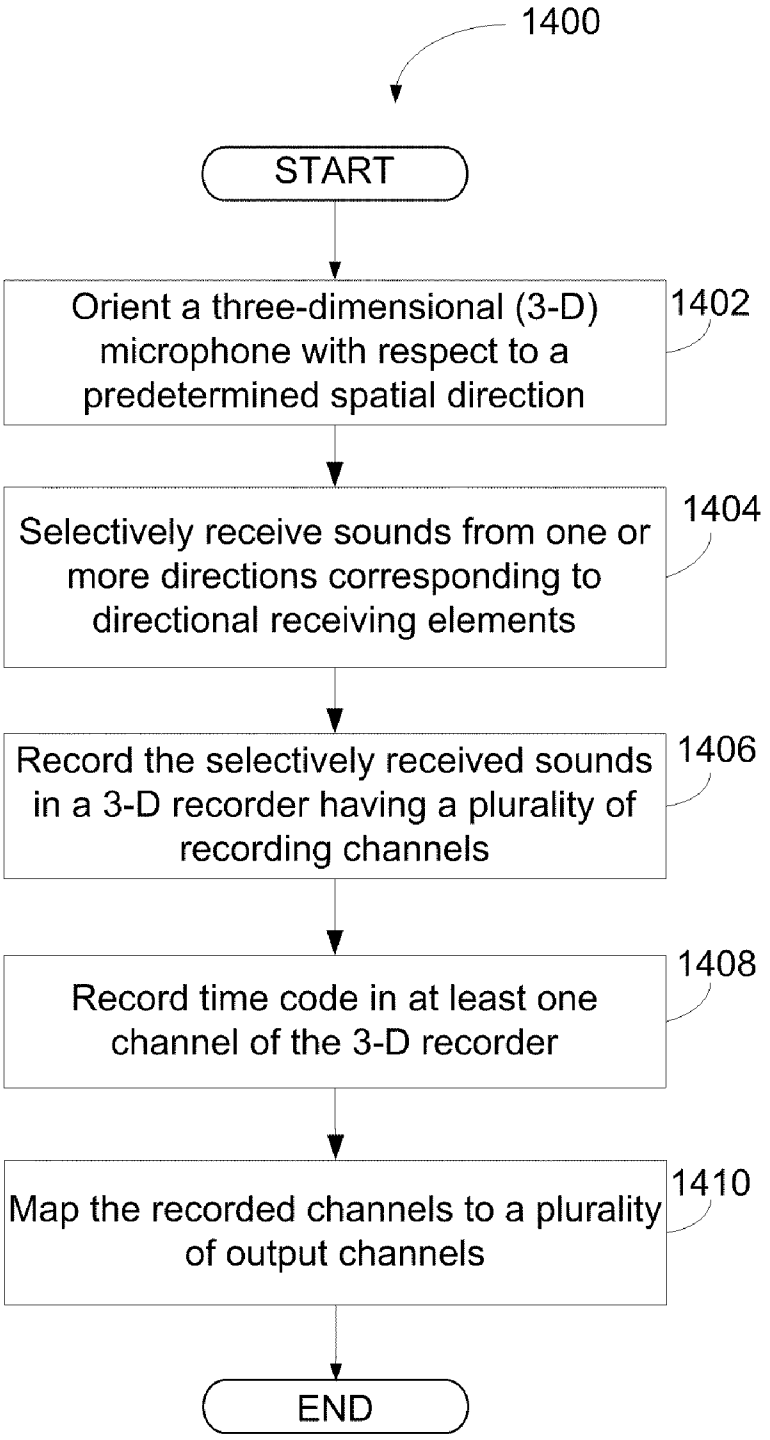

1400
START
Orient a three-dimensional (3-D) microphone with respect to a predetermined spatial direction
1402
Selectively receive sounds from one or more directions corresponding to directional receiving elements
1404
Record the selectively received sounds in a 3-D recorder having a plurality of recording channels
1406
Record time code in at least one channel of the 3-D recorder
1408
Map the recorded channels to a plurality of output channels
1410
END

FIG.14

SYSTEMS, METHODS, AND APPARATUS FOR CALIBRATING SPEAKERS FOR THREE-DIMENSIONAL ACOUSTICAL REPRODUCTION

CROSS-REFERENCE TO RELATED APPLICATION

[0001] This application claims benefit of U.S. Provisional Application No. 61/169,044, filed Apr. 14, 2009, which is incorporated herein by reference in its entirety.

RELATED APPLICATIONS

[0002] This application is related to application Ser. No. _____, filed concurrently with the present application on _____, entitled: "Systems, Methods, and Apparatus for Controlling Sounds in a Three-Dimensional Listening Environment," the contents of which are hereby incorporated by reference in their entirety.

[0003] This application is also related to application Ser. No. _____, filed concurrently with the present application on _____, entitled: "Systems, Methods, and Apparatus for Recording Multi-Dimensional Audio," the contents of which are hereby incorporated by reference in their entirety.

FIELD OF THE INVENTION

[0004] The invention generally relates to sound audio processing, and more particularly, to systems, methods, and apparatus for calibrating speakers for three dimensional acoustical reproduction.

BACKGROUND OF THE INVENTION

[0005] The terms "multi-channel audio" or "surround sound" generally refer to systems that can produce sounds that appear to originate from multiple directions around a listener. With the recent proliferation of computer games and game consoles, such as the Microsoft® X-Box®, the PlayStation®3 and the various Nintendo®-type systems, combined with at least one game designer's goal of "complete immersion" in the game, there exists a need for audio systems and methods that can assist the "immersion" by encoding three dimensional (3-D) spatial information in a multi-channel audio recording. The conventional and commercially available systems and techniques including Dolby Digital, DTS, and Sony Dynamic Digital Sound (SDDS) may be used to reproduce sound in the horizontal plane (azimuth), but such conventional systems may not adequately reproduce sound effects in elevation to recreate the experience of sounds coming from overhead or under-foot. Therefore, a need exists for systems and methods to record, decode, process and accurately reproduce 3-D sounds for a listening environment and for use with gaming consoles or other sources of visual information. Furthermore, there exists a need for initializing or calibrating a multi-speaker system for use with a 3-D audio system.

SUMMARY OF THE INVENTION

[0006] Embodiments of the invention can address some or all of the needs described above. According to embodiments of the invention, disclosed are systems, methods, and apparatus for calibrating speakers for three dimensional acoustical reproduction.

[0007] According to an example embodiment of the invention, a method is provided for initializing a plurality of speakers in a 3D acoustical reproduction system. The method includes positioning one or more calibration microphones near a listener position, generating calibration tones, selectively routing calibration tones to one or more of the plurality of speakers, producing audible tones from the plurality of speakers based on the generated calibration tones, sensing audible tones from the plurality of speakers with the one or more calibration microphones, determining one or more parameters associated with the plurality of speakers based on sensing the audible tones, and modifying settings of the 3D acoustical reproduction system based on the one or more determined parameters.

[0008] According to an example embodiment of the invention, a system is provided for initializing a plurality of speakers. The system includes a plurality of speakers, at least one calibration tone generator, one or more calibration microphones for sensing audible tones from the plurality of speakers, at least one memory, and one or more processors in communication with the memory. The one or more processors are configured for selectively generating and routing calibration tones to one or more of the plurality of speakers for producing audible tones, determining one or more parameters associated with the plurality of speakers based on sensing the audible tones, and modifying settings of the 3D acoustical reproduction system based on the one or more determined parameters.

[0009] According to an example embodiment of the invention, an apparatus is provided for producing multi-dimensional audio. The apparatus can include one or more input interfaces for receiving signals associated with multi-dimensional audio, one or more output interfaces, and a processor in communication with the one or more input interfaces and the one or more output interfaces. The processor includes at least one analog to digital converter, a signal router, and six or more programmable preamps.

[0010] Other embodiments and aspects of the invention are described in detail herein and are considered a part of the claimed invention. Other embodiments and aspects can be understood with reference to the following detailed description, accompanying drawings, and claims.

BRIEF DESCRIPTION OF THE DRAWINGS

[0011] Reference will now be made to the accompanying tables and drawings, which are not necessarily drawn to scale, and wherein:

[0012] FIG. **1** depicts an example system block diagram in accordance with an embodiment of invention.

[0013] FIG. **2** illustrates an example speaker perspective arrangement for a system in accordance with an embodiment of the invention.

[0014] FIG. **3** illustrates an example speaker placement top-down view, in accordance with an embodiment of the invention.

[0015] FIG. **4** illustrates an example 3D-EA system for recording 3-D audio, in accordance with an example embodiment of the invention.

[0016] FIG. **5** illustrates an example system for converting, routing, and processing audio and video, in accordance with an example embodiment of the invention.

[0017] FIG. **6** illustrates an example 3D-EA sound localization map, according to an example embodiment of the invention.

[0018] FIG. **7** illustrates an example look-up table of relative speaker volume levels for placement of sound at the 3D-EA localization regions of FIG. **6**.

[0019] FIG. **8** illustrates an example method flow chart for recording and encoding 3-D audio and optional video timecode in accordance with an embodiment of the invention.

[0020] FIG. **9** illustrates an example method flow chart for calibrating speakers, according to an example embodiment of the invention.

[0021] FIG. **10** illustrates an example method flow chart for converting, routing, processing audio and video in accordance with example embodiments of the invention.

[0022] FIG. **11** illustrates an example method flow chart for utilizing 3-D headphones in accordance with example embodiments of the invention.

[0023] FIG. **12** illustrates another example method flow chart for initializing and/or calibrating speakers, according to an example embodiment of the invention.

[0024] FIG. **13** illustrates an example method flow chart for controlling the placement of sounds in a three-dimensional listening environment.

[0025] FIG. **14** illustrates another example method flow chart for recording multi-dimensional audio, according to an example embodiment of the invention.

DETAILED DESCRIPTION OF EMBODIMENTS

[0026] Embodiments of the invention will now be described more fully hereinafter with reference to the accompanying drawings. This invention may, however, be embodied in many different forms and should not be construed as limited to the embodiment set forth herein; rather, embodiments are provided so that this disclosure will be thorough and complete, and will convey the scope of the invention.

[0027] FIG. **1** depicts an example system **100** in accordance with an embodiment of invention. The 3-D audio converter/amplifier **102** can accept and process audio from an external audio source **106**, which may include, for example, the audio output from a gaming console, the stereo audio from a standard CD player, tape deck, or other hi-fi stereo source, a mono audio source, or a digitized multi-channel source, such as Dolby 5.1 surround sound from a DVD player, or the like. The 3-D audio converter/amplifier **102** may also accept and process video from an external video source **104**, such as a gaming console, a DVD player, a video camera, or any source providing video information. The audio source **106** and video source **104** may be connected to separate input ports of the 3-D audio converter/amplifier, or the audio source **106** and video source **104** may be combined through one cable, such as HDMI, and the audio and video may be separated within the 3-D audio converter/amplifier **102** for further processing.

[0028] According to an example embodiment of the invention, the 3-D converter/amplifier **102** may provide both input and output jacks for example, to allow video to pass through for a convenient hook-up to a display screen. Detailed embodiments of the 3-D audio converter/amplifier **102** will be explained below in reference with FIG. **5**, but in general, the 3-D audio converter/amplifier **102** may provide processing, routing, splitting, filtering, converting, compressing, limiting, amplifying, attenuating, delaying, panning, phasing, mixing, sending, bypassing, etc., to produce, or re-produce 3D-EA sounds in a listening environment in both a horizontal plane (azimuth) and vertical plane (height) around the listener. According to an example embodiment, the 3-D audio converter/amplifier **102** may include an input for a video source **104**. The video source may be analyzed by the 3-D audio converter/amplifier **102**, either in real-time or near-real time, to extract spatial information that may be encoded or otherwise used for setting the parameters of the signals that may be sent to the speakers **110-120**, or to other external gear for further processing. In an example embodiment of the invention, the 3D-EA sound localization, or apparent directionality of the sonic information may be encoded and/or produced in relation to the position of objects within the 2-dimensional plane of a video image. Furthermore, according to an example embodiment of the invention, the 3D-EA sound localization may be automatically generated based at least in part on the processing and analysis of the video information, which may include relative depth information as well as information related to the position of objects within the 2-dimensional plane of the video image. According to an example embodiment, the system **100** may detect movement of an object in a video on the upper left side of the screen, and may shift the localization of the 3D-EA sound to the appropriate speakers to give the impression that the audio is coming from the upper left corner of the room, for example. According to other embodiments of the invention, object position information (provided either by automatic analysis of the video signal, or by object positional information encoded into the audio and relating to the video information) can be processed by the 3-D audio converter/amplifier **102** for dynamic positioning and/or placement of multiple 3D-EA sounds within a listening environment and optionally correlated with the positioning and/or placement of multiple objects in an associated video.

[0029] According to an example embodiment of the invention a speaker array, including speakers **110-120**, may be in communication with the 3-D audio converter/amplifier **102**, and may be responsive to the signals produced by the 3-D audio converter/amplifier **102**. In one embodiment, system **100** may also include a room calibration microphone **108**, as depicted in FIG. **1**. According to an example embodiment, the room calibration microphone **108** may contain one or more diaphragms for detecting sound simultaneously from one or more directions. The room calibration microphone **108** may be responsive to the time-varying sound pressure level signals produced by the speakers **110-120**, and may provide calibration input to the 3-D audio converter/amplifier **102** for proper setup of the various parameters (processing, routing, splitting, equalization, filtering, converting, compressing, limiting, amplifying, attenuating, delaying, panning, mixing, sending, bypassing, for example) within the 3-D audio converter/amplifier **102** to calibrate system **100** for a particular room. The room calibration microphone **108** may also be utilized in combination with a calibration tone generator within the 3-D audio converter/amplifier **102**, and speakers **110-120** appropriately placed in the listening environment, to automatically calibrate the system **100**. The details of this calibration procedure, in accordance with example embodiments of the invention will be discussed in the ROOM AND SPEAKER SETUP/CALIBRATION METHOD section below.

[0030] FIG. **2** illustrates an example speaker perspective arrangement for an example listening environment **200** for a 3D-EA system in accordance with an embodiment of the invention. According to an example embodiment the speakers, in communication with the 3-D audio converter/amplifier **102**, can be designated as Left **110**, Right **112**, Left Surround **114**, Right Surround **116**, Top Center Front **118**, and Top

Center Rear **120**. According to other example embodiments, the number and physical layout of speakers can vary within the environment **200**, and may also include a subwoofer (not shown). In accordance with an example embodiment of the invention, the Left **110**, Right **112**, Left Surround **114**, Right Surround **116**, speakers can be placed at ear level with respect to the listener position **202**. In accordance with another example embodiment of the invention, the Left **110**, Right **112**, Left Surround **114**, Right Surround **116**, speakers can be placed below ear level with respect to the listener position **202** to further extend the region of placement of 3D-EA sounds so that they appear to come from below the listener. In one example, an approximate equilateral triangle can be formed between the Left **110** speaker, the Right **112** speaker, and the listener position **202**. In another example, the Left **110** and Right **112** speakers can be oriented such that an acute angle of the isosceles triangle formed between the speakers **110**, **112** and the listener position **202** is between approximately 40 and approximately 60 degrees.

[0031] FIG. **2** also illustrates a Top Center Front speaker **118** and a Top Center Rear speaker **120** in accordance with an embodiment of the invention. These speakers **118**, **120** can respectively, be placed at front and rear of the listening environment **200**, vertically elevated above the listener position **202**, and can be angled downwards by approximately 10 to approximately 65 degrees to direct sound downwards towards the listener(s). The Top Center Front **118** speaker can be placed in the front of the environment **200** or room, typically above a viewing screen (not shown), and the Top Center Rear **120** speaker can be placed behind and above the listener position **202**. In this embodiment, the Top Center Rear **120** and Top Center Front **118** speakers may be pointed downwards at an angle towards the listener at listener position **202** so that the actual sonic reflections vibrate selective regions of cartilage within the ears of the listener to engage vertical or azimuth directional perception. According to an example embodiment of the invention, one or more of the speakers may be connected directly to the 3-D audio converter/amplifier **102** using two conductor speaker wires. According to another example embodiment of the invention, one or more of the speakers may be connected wirelessly to the 3-D audio converter/amplifier **102**.

[0032] Also depicted in FIG. **2** is the room calibration microphone **108**. As will be discussed further in the ROOM AND SPEAKER SETUP/CALIBRATION METHOD section below, the calibration microphone **108** may be wired or wireless, and may be in communication with the 3-D audio converter/amplifier **102**. According to an example embodiment, the calibration microphone, in cooperation with the 3-D audio converter/amplifier **102**, and speakers **110-120** may be utilized for any of the following: (a) to calibrate the speakers **110-120** for a particular room or listening environment **200**, (b) to aid in the setup and placement of the individual speakers for optimum 3D-EA performance, (c) to setup the equalization parameters for the individual channels and speakers, and/or (d) to utilize feedback to set the various parameters, speaker placements, etc.

[0033] FIG. **3** shows a top-down view of an example 3D-EA listening environment **300**, in accordance with an example embodiment of the invention. As measured with respect to the center line **306** bisecting the Top Center Front **118** and Top Center Rear **120** speakers, the Left **110** speaker may be centered on line **308** extending from position of the listener to form an angle **304** with the center line **306** of approximately 30 degrees. Depending on the room configuration and other factors related to the optimum 3D-EA sound, the angle **304** may range between about 10 and about 80 degrees. Similarly, the Left Surround speaker **114** may be centered on line **310** extending from position of the listener to form an angle **302** with the center line **306** of approximately 110 degrees. Depending on the room configuration and other physical limitations, or factors related to the optimum 3D-EA sound, the angle **304** may range between about 100 and about 160 degrees. The Right **112** and Right Surround **116** speakers may be placed in a mirror image with respect to the centerline **306** respectively with the Left **110** and Left Surround **114** speakers. As depicted in FIGS. **2** and **3**, the Top Center Front **118** and Top Center Rear **120** speakers may be placed on about the centerline (as their name suggest) and, as with the other speakers, may be pointed to direct 3D-EA sound towards the listener. According to example embodiments of the invention, the linear distance between the listener at listening position **202** (FIG. **2**), as depicted by the position of the calibration microphone **108** (FIG. **3**), and the individual speakers **110-120** may vary, and may depend on the room configuration, room physical limitations, factors related to the optimum 3D-EA sound, and size of 3D-EA listening sphere or dome **312** needed in order to reproduce 3D-EA sounds for one or more listeners. Typically, a 3D-EA listening sphere or dome **312** will have a radius smaller than the distance to the closest speaker **110-120**. However, according to an example embodiment of the invention, the size of the 3-D listening sphere or dome **310** may be expanded or contracted by selective processing, routing, volume control, and/or phase control of the driving energy directed to each of speakers **110-120**.

[0034] FIG. **4** depicts an example 3D-EA recording system **400**, according to an embodiment of the invention. The system **400** may be utilized to record and/or otherwise encode 3-D audio information from the source environment. According to an example embodiment, the 3D-EA recording system **400** may encode the naturally occurring directional information within a particular scene or environment to help minimize the manual processing of 3D-EA sounds that may otherwise be done during post production. According to an example embodiment, a binaural microphone system (not shown) may be utilized for recording audio. A typical binaural recording unit has two high-fidelity microphones mounted in a dummy head, and the microphones are inserted into ear-shaped molds to fully capture some or all of the audio frequency adjustments that can occur naturally as sound wraps around the human head and is "shaped" by the form of the outer and inner ear. According to another example embodiment, a 3-D microphone **410**, which may be similar to the calibration microphone **108** described above, may be utilized to selectively record sounds from multiple directions. According to an example embodiment, the 3-D microphone may have at least one diaphragm element per spatial dimension of directional sensitivity and encoding. The signals produced by the 3-D microphone **410** may be received and recorded via a 3-D sound recorder **402** having multiple input and storage channels. According to an example embodiment of the invention, the 3-D sound recorder **402** may simultaneously record time code **408** that is provided by a video camera **406**. According to an example embodiment of the invention, the 3-D sound recorder **402** may simultaneously record time code **408** that is provided by a time-code generator within the 3-D sound recorder **402**. After recording the

audio and time code, the information may be downloaded or otherwise transferred to an off-line sound processor **404** for further processing or storage. According to example embodiments of the invention, the audio and time code information may be further edited and processed for use with a video, an audio recording, or a computer game, for example.

[0035] FIG. **5** depicts a block diagram representation of the 3-D audio converter/amplifier, according to an example embodiment of the invention. Input terminals **504-510** can be utilized for receiving one or more input audio and/or video signal sources, including pre-processed 3D-EA. The input terminals **504-510** may include multiple input terminals (not shown) to facilitate a variety of source connections including, but not limited to, RCA, XLR, S/PDIF, digital audio, coaxial, optical, ¼" stereo or mono, ⅛" mini stereo or mono, DIN, HDMI and other types of standard connections. According to an example embodiment, the audio input terminals **504**, **506**, **508** may be in communication with an audio microprocessor **512**, and the video input terminal **510** may be in communication with a video microprocessor **538**. Each of the microprocessors **512**, **538** may be in communication with a memory device **550** and may either reside on the same or different integrated circuits.

[0036] According to an example embodiment of the invention, the audio microprocessor **512** may include a terminal select decoder A/D module **514**, which may receive signals from the input terminals **504-508**. The decoder **514** may be in communication with an input splitter/router **516**, which may be in communication with multi-channel leveling amplifiers **518**. The multi-channel leveling amplifiers **518** may be in communication with multi-channel filters/crossovers **520** which may be in communication with a multi-channel delay module **522**. The multi-channel delay module **522** may be in communication with multi-channel pre-amps **524**, which may be in communication with a multi-channel mixer **524**, which may be in communication with an output D/A converter **528**. The output of the audio microprocessor **512** may be in communication with multiple and selectable tube preamps **546**. The output from either the D/A converter **528**, or the tube preamps **546**, or a mix of both, may be in communication with multi-channel output amplifiers **530**, multiple tube output stages **548**, and a transmitter **548** for the wireless speakers. The output of the tube output stages **548** and/or the multi-channel output amplifiers **530**, or a mix of both may be in communication with output terminals **522**, which are further in communication with speakers. According to an example embodiment, the transmitter **548** for the wireless speakers may be in communication with a receiver associated with the wireless speaker (not shown). According to an example embodiment, a routing bus **542** and summing/mixing/routing nodes **544** may be utilized to route and connect all digital signals to-and-from any of the modules described above within the audio microprocessor **512**.

[0037] The 3-D audio converter/amplifier **102** may also include a touch screen display and controller **534** in communication with the audio microprocessor for controlling and displaying the various system settings. According to an example embodiment, the 3-D audio converter/amplifier **102** may include a wireless system for communication with the room calibration microphone **108** and a wireless remote control. A power supply **502** may provide power to all the circuits of the 3-D audio converter/amplifier **102**.

[0038] According to an example embodiment, the 3-D audio converter/amplifier **102** may include one or more input terminals **510** for video information. For example, one terminal may be dedicated to video information, while another is dedicated to video time code. The video input terminals **510** may be in communication with a video microprocessor **538** for spatial movement extraction. The video microprocessor **538** may be further in communication with the audio microprocessor **512**, and may provide spatial information for selectively processing the temporal audio information.

[0039] Again with reference to FIG. **5**, blocks of the audio microprocessor **512** within the 3-D audio converter/Amplifier will now be explained, according to example embodiments of the invention. The input terminal select decoder and A/D module **514** may selectively receive and transform the one or more input audio signals from the input terminals **504-508** (or from other input terminals not shown) as needed. According to an example embodiment, if information is present at the Optical/SPDIF terminal **504** in the form of a digital optical signal, the decoder **514** may detect the presence of the optical signal, and may perform the appropriate switching and optical to electrical conversion. According to example embodiments of the invention, the decoder **514** may automatically select input terminals via a signal detection process, or it may require manual input by the user, particularly in the case where multiple input signals may be present, and when one particular input is desired. According to example embodiments of the invention, the terminal select decoder and A/D module **514** may include additional sub-modules for performing terminal sensing, terminal switching, transformations between optical and electrical signals, sensing the format of the digital or analog signal, and performing transformations from analog to digital signals. According to an example embodiment, analog audio signals may be converted to digital signals via an A/D converter within the terminal select decoder A/D module **514**, and as such, may remain in digital format until converted back to analog at the D/A module **528** prior to being amplified and sent to the speakers. Conversely, digital signals present on the input terminals may bypass the A/D sub module processing since they are already in the digital format. The signal flow in FIG. **5** indicates digital signals as dashed lines, according to an example embodiment of the invention, however, according to other example embodiments of the invention, input signals (analog or digital) may be routed to bypass one or more of the modules **516-528**, and yet in other embodiments of the invention, one or more of the modules **514-528** may include the capability to process either digital or analog information.

[0040] With continued reference to FIG. **5**, and according to an example embodiment of the invention, a multi-signal bus **542** with multiple summing/mixing/routing nodes **544** may be utilized for routing, directing, summing, mixing, signals to and from any of the modules **514-528**, and/or the calibration tone generator **540**. According to an example embodiment, the input splitter/router module **516** may receive digital signals from decoder **514**, and may act as an input mixer/router for audio signals, either alone, or in combination with the bus **542** and the summing/mixing/routing nodes **544**. The input splitter/router module **516** may also receive a signal from the calibration tone generator **540** for proper routing through the rest of the system. According to an example embodiment of the invention, the input splitter/router module **516** may perform the initial audio bus **542** input routings for the audio microprocessor **512**, and as such, may be in parallel communication with the downstream modules, which will be briefly described next.

[0041] According to an example embodiment of the invention, the audio microprocessor **512** may include multi-channel leveling amplifiers **518** that may be utilized to normalize the incoming audio channels, or to otherwise selectively boost or attenuate certain bus **542** signals. According to an example embodiment, the leveling amps **518** may precede the input splitter/router **516**. According to an example embodiment, the leveling amps **518** may be in parallel communication with any of the modules **520-528** and **540** via a parallel audio bus **542** and summing/mixing/routing nodes **544**. According to an example embodiment, the audio microprocessor **512** may also include a multi-channel filter/crossover module **520** that may be utilized for selective equalization of the audio signals. According to an example embodiment, one function of the multi-channel filter/crossover module **520** may be to selectively alter the frequency content of certain audio channels so that, for example, only relatively mid and high frequency information is directed to the Top Center Front **118** and Top Center Rear **120** speakers, or so that only the low frequency content from all channels is directed to a subwoofer speaker.

[0042] With continued reference to FIG. **5**, and according to an example embodiment, the audio microprocessor **512** may include a multi-channel delay module **522**, which may receive signals from upstream modules **514-520**, **540**, in any combination via a parallel audio bus **542** and summing/mixing/routing nodes **544**, or by the input splitter router **516**. The multi-channel delay module **522** may be operable to impart a variable delay to the individual channels of audio that may ultimately be sent to the speakers. The multi-channel delay module **522** may also include a sub-module that may impart phase delays, for example, to selectively add constructive or destructive interference within the 3D-EA listening sphere or dome **312**, or to adjust the size and position of the 3D-EA listening sphere or dome **312**.

[0043] According to an example embodiment of the invention, the audio microprocessor **512** may further include a multi-channel-preamp with rapid level control **524**. This module **524** may be in parallel communication with all of the other modules in the audio microprocessor **512** via a parallel audio bus **542** and summing/mixing/routing nodes **544**, and may be controlled, at least in part, by the encoded 3-D information, either present within the audio signal, or by the 3-D sound localization information that is decoded from the video feed via video microprocessor **538**. An example function provided by the multi-channel-preamp with rapid level control **524** may be to selectively adjust the volume of one or more channels so that the 3D-EA sound may appear to be directed from a particular direction. According to an example embodiment of the invention, a mixer **526** may perform the final combination of the upstream signals, and may perform the appropriate output routing for directing a particular channel. The mixer **526** may be followed by a multiple channel D/A converter **528** for reconverting all digital signals to analog before they are further routed. According to one example embodiment, the output signals from the D/A **528** may be optionally amplified by the tube pre-amps **546** and routed to transmitter **548** for sending to wireless speakers. According to another example embodiment, the output from the D/A **528** may be amplified by one or more combinations of (a) the tube pre-amps **546**, (b) the multi-channel output amplifiers **530**, or (c) the tube output stages **548** before being directed to the output terminals **532** for connecting to the speakers. According to an example embodiment of the invention, the multi-channel output amplifiers **530** and the tube output stages **548** may include protection devices to minimize any damage to speakers hooked to the output terminals **523**, or to protect the amplifiers **530** and tube output stages **548** from damaged or shorted speakers, or shorted terminals **532**.

[0044] According to an example embodiment certain 3D-EA output audio signals can be routed to the output terminals **532** for further processing and/or computer interfacing. In certain instances, an output terminal **532** may include various types of home and/or professional quality outputs including, but not limited to, XLR, AESI, Optical, USB, Firewire, RCA, HDMI, quick-release or terminal locking speaker cable connectors, Neutrik Speakon connectors, etc.

[0045] According to example embodiments of the invention, speakers for use in the 3-D audio playback system may be calibrated or initialized for a particular listening environment as part a setup procedure. The setup procedure may include the use of one or more calibration microphones **536**. In an example embodiment of the invention, one or more calibration microphones **536** may be placed within about 10 cm of a listener position. In an example embodiment, calibration tones may be generated and directed through speakers, and detected with the one or more calibration microphones **536**. In certain embodiments of the invention, the calibration tones may be generated, selectively directed through speakers, and detected. In certain embodiments, the calibration tones can include one or more of impulses, chirps, white noise, pink noise, tone warbling, modulated tones, phase shifted tones, multiple tones or audible prompts.

[0046] According to example embodiments, the calibration tones may be selectively routed individually or in combination to a plurality of speakers. According to example embodiments, the calibration tones may be amplified for driving the speakers. According to example embodiments of the invention, one or more parameters may be determined by selectively routing calibration tones through the plurality of speakers and detecting the calibration tones with the calibration microphone **536**. For example, the parameters may include one or more of phase, delay, frequency response, impulse response, distance from the one or more calibration microphones, position with respect to the one or more calibration microphones, speaker axial angle, speaker radial angle, or speaker azimuth angle. In accordance with an example embodiment of the invention, one or more settings, including volume, equalization, and/or delay, may be modified in each of the speakers associated with the 3D-EA system based on the calibration or setup process. In accordance with embodiments of the invention, the modified settings or calibration parameters may be stored in memory **550**. In accordance with an example embodiment of the invention, the calibration parameters may be retrieved from memory **550** and utilized to automatically initialize the speakers upon subsequent use of the system after initial setup.

Sound Localization

[0047] FIG. **6** depicts a 3D-EA sound localization map **600**, according to an example embodiment of the invention. The 3D-EA sound localization map **600** may serve as an aid for describing, in space, the relative placement of the 3D-EA sound localizations relative to a central location. According to an example embodiment, the 3D-EA sound localization map **600** may include three vertical levels, each with 9 sub-regions, for a total of 27 sub-regions placed in three dimensions around the center sub-region **14**. The top level may consist of

sub-regions **1-9**; the middle level may consist of sub-regions **10-18**; and the bottom level may consist of sub-regions **19-27**. An example orientation of a listening environment may place the center sub-region **14** at the head of the listener. The listener may face forward to look directly at the front center sub-region **11**. According to other embodiments, the 3D-EA sound localization map may include more or less sub-regions, but for the purposes of defining general directions, vectors, localization, etc. of the sonic information, the 3D-EA sound localization map **600** may provide a convenient 3-D framework for the invention. As discussed in the preceding paragraphs, and in particular, with respect to FIG. **5**, one aspect of the 3-D audio converter/amplifier **102** is to adjust, in real or near-real time, the parameters of the multiple audio channels so that all or part of the 3D-EA sound is dynamically localized to a particular region in three dimensional space. According to other example embodiments, the 3D-EA sound localization map **600** may include more or less sub-regions. According to another example embodiment, the 3D-EA sound localization map **600** may have a center offset vertically with respect to the center region shown in FIG. **6**. The 3D-EA sound localization map **600** may be further explained and defined in terms of audio levels sent each speaker to localize 3D-EA sound at any one of the sub-regions **1-27** with the aid of FIG. **7**.

[0048] According to an example embodiment of the invention, FIG. **7** depicts a example look-up table of relative sound volume levels (in decibels) that may be set for localizing the 3D-EA sound near any of the 27 sub-regions. The symbols "+", "−", "0", and "off" represent the relative signal levels for each speaker that will localize the 3D-EA sound to one of the 27 sub-regions, as shown in FIG. **6**. According to an example embodiment of the invention, the "0" symbol may represent the default level for a particular speaker's volume, which may vary from speaker to speaker. For example, the Top Center Front **118** and Top Center Rear **120** speakers may have a default "0" level that is about 6 dB less than the default level "0" for the left speaker **110**. According to an example embodiment of the invention, the "+" symbol may represent +6 dB, or approximately a doubling of the volume with respect to the default "0" signal level. The "−" symbol may represent about −6 dB, or approximately one half of the volume with respect to the default "0" level of the signal. The symbol "off" indicates that there should be no signal going to that particular speaker. In other example embodiments, the "+" symbol may represent a range of levels from approximately +1 to approximately +20 dB, depending on factors such as the size of the 3D-EA listening sphere or dome **312** needed in order to reproduce 3D-EA sounds for one or more listeners. Likewise, the "−" symbol may represent a range of levels from approximately −1 to approximately −20 dB. According to an example embodiment of the invention, the size of the 3D-EA listening sphere or dome **312** may be expanded or compressed by value of the signal level assigned to the "+" and "−" symbols.

[0049] In accordance with example embodiments of the invention, signals may be adjusted to control the apparent localization of sounds in a 3-dimensional listening environment. In an example embodiment, audio signals may be selectively processed by adjusting one or more of delay, equalization, and/or volume. In an example embodiment the audio signals may be selectively processed based on receiving decode data associated with the one or more audio channels. In accordance with an example embodiment, the decode data may include routing data for directing specific sounds to specific speakers, or to move sounds from one speaker (or set of speakers) to another to emulate movement. According to example embodiments, routing the one or more audio channels to one or more speakers may be based at least in part on the routing data. In certain embodiments, routing may include amplifying, duplicating and/or splitting one or more audio channels. In an example embodiment, routing may include directing the one or more audio channels to six or more processing channels. In certain embodiments, the audio may be processed for placing sounds in any one of 5 or more apparent locations in the 3-dimensional listening environment.

3-D Sound Recording Method

[0050] The method for recording 3-D audio, according to an example embodiment of the invention, will now be described with respect to FIG. **4** and the flowchart of FIG. **8**. Method **800** begins in block **802** where a 3-D microphone **410** is connected to a multi-channel recorder **402**. The 3-D microphone **410** may have multiple diaphragms or elements, each with a directional sensitivity that may selectively detect sonic information from a particular direction, depending on the orientation of the element. The directional receiving elements or diaphragms may comprise condenser elements, dynamic elements, crystal elements, piezoelectric elements, or the like. The diaphragms may have a cardioid, or super-cardioid sensitivity patterns, and may be oriented with respect to their nearest neighbors for partial overlap of their acceptance or sensitivity patterns. The 3-D microphone **410** may have 3 or more diaphragms for partial 3-D or whole sphere coverage. The 3-D microphone **410** may have an indicator or marking for proper directional orientation within a particular space.

[0051] Method **800** continues in optional block **804** where time code **408** from a video camera **406** (or other time code generating equipment) may be input to the 3-D recorder **402**, recorded in a separate channel, and used for playback synchronization at a later time. Optionally, the 3-D recorder **402** may include an internal time code generator (not shown).

[0052] Method **800** continues in optional block **805** where parallax information from a stereo camera system **412** may be utilized for detecting the depth information of an object. The parallax information associated with the object may further be utilized for encoding the relative sonic spatial position, direction, and/or movement of the audio associated with the object.

[0053] The method continues in block **806** where the 3-D audio information (and the time code) may be recorded in a multi-channel recorder **402**. The multi-channel 3-D sound recorder **402** may include microphone pre-amps, automatic gain control (AGC), analog-to-digital converters, and digital storage, such as a hard drive or flash memory. The automatic gain control may be a linked AGC where the gain and attenuation of all channels can be adjusted based upon input from one of the microphone diaphragms. This type of linked AGC, or LAGC, may preserve the sonic spatial information, limit the loudest sounds to within the dynamic range of the recorder, and boost quiet sounds that may otherwise be inaudible.

[0054] Method **800** continues in block **808** with the processing of the recorded 3-D audio information. The processing of the 3-D audio information may be handled on-line, or optionally be transferred to an external computer or storage device **404** for off-line processing. According to an example embodiment of the invention, the processing of the 3-D audio

information may include analysis of the audio signal to extract the directional information. As an illustrative example, suppose 3-D recorder is being used to record a scene of two people talking next to road, with the microphone positioned between the road and the people. Presumably, all of the microphone channels will pick up the conversation, however the channels associated with the diaphragms closest to the people talking will likely have larger amplitude signal levels, and as such, may provide directional information for the conversation relative to the position of the microphone. Now, assume that a car travels down the street. As the car travels, the sound may be predominant in one channel associated with the microphone diaphragm pointed towards the car, but the predominant signal may move from channel to channel, again providing directional information for the position of the car with respect to time. According to an example embodiment of the invention, the multiple-diaphragm information, as described above, may be used to encode directional information in the multi-channel audio. Method **800** ends after block **810** where the processed 3-D information may be encoded into the multiple audio channels.

[0055] Another method for recording multi-dimensional audio is discussed with reference to FIG. **14** below.

[0056] According to one example embodiment of the invention, the signals recorded using the 3-D microphone may be of sufficient quality, with adequate natural directionality that no further processing is required. However, according to another example embodiment, the 3-D microphone may have more or fewer diaphragms than the number of speakers in the intended playback system, and therefore, the audio channels may be mapped to channels corresponding with the intended speaker layout. Furthermore, in situations requiring conventional recording techniques using high quality specialized microphones, the 3-D microphone may be utilized primarily for extracting 3D-EA sonic directional information. Such information may be used to encode directional information onto other channels that may have been recorded without the 3-D microphone. In some situations, the processing of the 3-D sound information may warrant manual input when sonic directionality can not be determined by the 3-D microphone signals alone. Other situations are envisioned where it is desirable to encode directional information into the multi-channel audio based on relative position of an object or person within a video frame. Therefore, the method of processing and encoding includes provisions for manual or automatic processing of the multi-channel audio.

[0057] According to certain embodiments of the invention, sounds emanating from different directions in a recording environment may be captured and recorded using a 3-D microphone having multiple receiving elements, where each receiving element may be oriented to preferentially capture sound coming predominately from a certain direction relative to the orientation of the 3-D microphone. According to example embodiments, the 3-D microphone may include three or more directional receiving elements, and each of the elements may be oriented to receive sound coming from a predetermined spatial direction. In accordance with embodiments of the invention, sounds selectively received buy the directions receiving elements may be recorded in separate recording channels of a 3-D sound recorder.

[0058] According to an example embodiment, the 3-D recorder may record time code in at least one channel. In one embodiment, the time code may include SMTPE, or other industry standard formats. In another embodiment, the time code may include relative time stamp information that can allow synchronization with other devices. According to an example embodiment, time code may be recorded in at least one channel of the 3-D recorder, and the time code may be associated with at least one video camera.

[0059] According to example embodiments of the invention, the channels recorded by the 3-D recorder may be mapped or directed to output paths corresponding to a predetermined speaker layout. In certain embodiments, the recorded channels may be mapped or directed to output paths corresponding to six speakers. In certain example embodiments, recorded channels may be directed to output channels that correspond to relative position of an object within a video frame.

Room and Speaker Setup/Calibration Method

[0060] FIG. **9** depicts a method **900** for setting-up and calibrating a 3-D audio system **100**, according to an example embodiment of the invention. Beginning at block **902**, the calibration microphone **108** may be connected to the 3-D audio converter/amplifier, either wirelessly, or wired. According to an example embodiment of the invention, the calibration microphone **108** may include one or more directionally sensitive diaphragms, and as such, may be similar or identical to the 3-D microphone **410** described above. The method continues in block **904** where the speakers **110-120** are connected to corresponding output terminals **532**. Optionally, if one or more of the speakers are wireless, they can be in communication with the transmitter **548** for the wireless speakers. The setup mode of the 3-D audio converter/amplifier power may be entered manually, or automatically based upon the presence of the calibration microphone. The setup/calibration method continues in block **906** where, according to an example embodiment of the invention, the calibration microphone may measure the relative phase and amplitude of special tones generated by the calibration tone generator **540** within the 3-D audio converter amplifier and output through the speakers **110-120**. The tones produced by the calibration tone generator **540** may include impulses, chirps, white noise, pink noise, tone warbling, modulated tones, phase shifted tones, and multiple tones, and may be generated in an automatic program where audible prompts may be given instructing the user to adjust the speaker placement or calibration microphone placement.

[0061] Method **900** continues in block **908** where according to an example embodiment of the invention, signals measured by the calibration microphone **106** may be used as feedback for setting the parameters of the system **100**, including filtering, delay, amplitude, and routing, etc for normalizing the room and speaker acoustics. The method continues at block **910** where the calibration process can be looped back to block **906** to setup additional parameters, remaining speakers, or placement of the calibration microphone **106**. Looping though the calibration procedure may be accompanied by audible or visible prompts, for example "Move the calibration microphone approximately 2 feet to the left, then press enter." so that the system can properly setup the 3D-EA listening sphere or dome **312**. Otherwise, if all of the calibration procedure has completed, the method may continue to block **912** where the various calibration parameters calculated during the calibration process may be stored in non-volatile memory **550** for automatic recall and setup each time the system is subsequently powered-on so that calibration is necessary only when the system is first setup in a room, or when the user

desires to modify the diameter of the 3D-EA listening sphere or dome **312**, or when other specialized parameters are setup in accordance with other embodiments of the invention. The method **900** ends at block **914**.

[0062] An additional method for initializing and/or calibrating speakers associated with the 3D-EA system will be further described below with reference to FIG. **12**.

[0063] According to an example embodiment of the invention, a method **1000** is shown in FIG. **10** for utilizing the 3-D audio converter/amplifier for playback. Starting at block **1002**, the input devices (audio source, video source) may be hooked to the input terminals of the 3-D audio converter/amplifier **102**. Next, in block **1003**, the system can be optionally calibrated, as was described above with reference to the flowchart of FIG. **9**. For example, if the system was previously calibrated for the room, then the various pre-calculated parameters may be read from non-volatile memory **550**, and calibration may not be necessary. The method **1000** continues in block **1004** where the input terminals are selected, either manually, or automatically by detecting the input on terminals. The method **1000** may then continue to decision block **1006** where a determination can be made as to the decoding of the audio. If the terminal select decoder A/D **514** module detects that the selected input audio is encoded, it may decode the audio, as indicated in block **1008**. According to an example embodiment, the decoding in block **1008** may, for example, involve splitting a serial data stream into several parallel channels for separate routing and processing. After decoding, the terminal select decoder A/D **514** module may also be used to convert analog signals to digital signals in block **1010**, however this A/D block may be bypassed if the decoded signals are already in digital format. If, in decision block **1006**, the audio is determined to be generic analog stereo audio with no encoding, then the method may proceed to block **1012** where the analog signal may be converted to digital via a multi-channel A/D converter. According to an example embodiment, the method from either block **1010** or block **1012** may proceed to block **1016** where routing functions may be controlled by the input splitter/router module **516** in combination with the multi-channel bus **542** and the summing/mixing/routing nodes **544**. According to multiple example embodiments of the invention, after block **1016**, any number of unique combinations of routing and combining of the signals may be provided by the audio microprocessor **512**. The routing and combining may involve processing of the digital signals from any, all, or none of blocks **1018-1026**. For example, the multiple channels of audio may all be routed through the leveling amps **518** and the multi channel preamps with rapid level control **514**, but some of the channels may also be routed through the crossovers **520** and/or the delay module **522**. In other example embodiments, all channels may be routed through all of the modules **518-526** (corresponding to blocks **1018-2026** in FIG. **10**), but only certain channels may be processed by the modules.

[0064] According to an example embodiment of the invention, block **1014** depicts video information that may be utilized for dynamic setting of the parameters in the corresponding blocks **1018-1026**. For example, the video information in block **1014** may be utilized to interact with the level control in block **1024** (corresponding to the rapid level control **524** in FIG. **5**) to rapidly adjust the relative volume levels of each channel to dynamically place certain sounds within a subregion of the 3D-EA listening sphere or dome **312**, as was discussed in relation with FIGS. **6** and **7**. In another example embodiment, the video information in block **1014** may be utilized to interact with other blocks, such as the delay block **1020** and/or the filtering/crossover block **1022** to control the apparent location of a 3D-EA sound by imparting phasing or by adjusting the frequency content of a sound in certain speakers relative to the phasing or frequency content of the other speakers.

[0065] After the processing of the signals, the method **1000** continues to D/A block **1028** where the digital signals may be converted to analog before further routing. The method may continue to block **1030** where the analog signals can be pre-amplified by either a tube pre-amp, a solid state preamp, or a mix of solid state and tube preamps. According to one example embodiment, the output preamp of block **1030** may also be bypassed. The pre-amplified or bypassed signal may then continue to one or more paths as depicted in block **1032**. In one example embodiment, the signals may be output amplified by multi-channel output amplifiers **530** before being sent to the output terminals. According to an example embodiment, multi-channel output amplifiers may include 6 or more power amplifiers. According to another example embodiment, the signals may be output amplified by tube output stages **548** before being routed to the output terminals. In yet another example embodiment, the signals may be sent to a multi-channel wireless transmitter **548** for transmitting to wireless speakers. In this embodiment, line-level signals can be sent to the wireless transmitter, and the warmth of the tube preamps **546** may still be utilized for the signals routed to separate amplifiers in the wireless speakers. According to another example embodiment, and with reference to block **1032**, any combination of the output paths described above can be provided including wireless, tube output, solid state output, and mix of the wireless, tube, and solid state outputs. The method of FIG. **10** ends at block **1034**, but it should be apparent that the method is dynamic and may continuously repeat, particularly from block **1016** to block **1028** as the system operates.

[0066] An additional method for controlling the apparent localization of sounds in a 3-dimensional listening environment will be further described below with reference to FIG. **13**.

3-D Headphones

[0067] According to an example embodiment of the invention, the speakers or transducers utilized in the 3D-EA reproduction, may be mounted within headphones, and may be in communication with the 3-D Audio Converter/Amplifier **102** via one or more wired or wireless connections. According to an example embodiment of the invention, the 3-D headphones (not shown) may include at least one orientation sensor (accelerometer, gyroscope, weighted joystick, compass, etc.) to provide orientation information that can be used for additional dynamic routing of audio signals to the speakers within the 3-D headphones. According to an example embodiment, the dynamic routing based on the 3-D headphone orientation may be processed via the 3-D Audio Converter/Amplifier **102**. According to another example embodiment, the dynamic routing based on the 3-D headphone orientation may be processed via additional circuitry, which may include circuitry residing entirely within the headphones, or may include a separate processing box for interfacing with the 3-D Audio Converter/Amplifier **102**, or for interfacing with other audio sources. Such dynamic routing can simulate a virtual listening environment where the rela-

tive direction of 3D-EA sounds can be based upon, and may correspond with the movement and orientation of the listener's head.

[0068] An example method **1100** for providing dynamic 3D-EA signal routing to 3-D headphones based on the listener's relative orientation is shown in FIG. **11**. The method begins in block **1102** where the 3-D headphones may be connected to the 3-D audio converter/amplifier **102** via one or more wired or wireless connections. For example, the wireless connections for transmitting orientation information to the 3-D audio converter/amplifier **102** may include the wireless link associated with the remote control or the calibration mic **536**, as shown in FIG. **5**. The wireless information for transmitting audio signals from the 3-D audio converter/amplifier **102** to the 3-D headphones may include the transmitter for wireless speakers **548**. According to another embodiment, a multi-conductor output jack may be included in the output terminals **532** to provide amplified audio to the headphones so that separate amplifiers may not be required.

[0069] The method continues in block **1104** where, according to an example embodiment of the invention, the nominal position of the orientation sensor may be established so that, for example, any rotation of the head with respect to the nominal position may result in a corresponding rotation of the 3D-EA sound field produced by the 3-D headphones. In an example embodiment, the listener may establish the nominal position by either pressing a button on the 3-D headphones, or by pressing a button on the remote control associated with the 3-D audio converter/amplifier **102** to establish the baseline nominal orientation. In either example case, the 3-D headphone processor (either in the 3-D audio converter/amplifier **102**, in the 3-D headphones themselves, or in an external processor box) may take an initial reading of the orientation sensor signal when the button is pressed, and may use the initial reading for subtracting, or otherwise, differentiating subsequent orientation signals from the initial reading to control the 3D-EA sound field orientation.

[0070] The method continues in block **1106** where, according to an example embodiment, signals from the one or more orientation sensors may be transmitted to the 3-D audio converter/amplifier **102** for processing the 3D-EA sound field orientation. As described above, the signal from the orientation sensor may reach the 3-D audio converter/amplifier **102** via a wired or wireless connection. According to another example embodiment, the signals from the one or more orientation sensors may be in communication with the 3-D headphone processor, and such a processor may reside within the 3-D audio converter/amplifier **102**, within the 3D headphones, or within a separate processing box.

[0071] The method continues in block **1108** where, according to an example embodiment of the invention, the signals from the one or more orientation sensors may be used to dynamically control and route the 3-D audio output signals to the appropriate headphone speakers to correspond with head movements. The method ends at block **1110**.

[0072] It should be apparent from the foregoing descriptions that all of the additional routing and processing of the signals for the 3-D headphones may be done in addition to the routing and processing of the audio signals for placement of 3D-EA sounds within a 3D-EA listening sphere or dome **312**. For example, a sound coming from the direct left, which may be region **13** as shown in FIG. **6**, may be rotated to the right to the position of region **11** as the listener's head rotates 90 degrees to the left about the vertical axis. Therefore, in an example embodiment, the 3D-EA sound-field within the headphones may rotate in a direction opposing the rotation of the listener's head.

Remote Operations

[0073] According to example embodiments of the invention, the 3-D audio converter/amplifier **102** may include one or more remote control receivers, transmitters, and/or transceivers for communicating wirelessly with one or more remote controls, one or more wireless microphones, and one or more wireless or remote speakers or speaker receiver and amplification modules. In an example embodiment, the wireless or remote speaker receiver and amplification modules can receive 3D-EA signals from a wireless transmitter **548**, which may include capabilities for radio frequency transmission, such as Bluetooth. In another example embodiment the wireless transmitter **548** may include infrared (optical) transmission capabilities for communication with a wireless speaker or module. In yet another example embodiment, the power supply **502** may include a transmitter, such as an X10 module **552**, in communication with the output D/A converter **528** or the tube pre-amp **546**, for utilizing existing power wiring in the room or facility for sending audio signals to remote speakers, which may have a corresponding X10 receiver and amplifier.

[0074] In an example embodiment, a wireless or wired remote control may be in communication with the 3-D audio converter/amplifier **102**. In an example embodiment, the a wireless or wired remote control may communicate with the 3-D audio converter/amplifier **102** to, for example, setup speaker calibrations, adjust volumes, setup the equalization of the 3D-EA sound in the room, select audio sources, or to select playback modes. In another example embodiment, the wireless or wired remote control may communicate with the 3-D audio converter/amplifier **102** to setup a room expander feature, or to adjust the size of the 3D-EA listening sphere or dome **312**. In another example embodiment, the wireless or wired remote control may comprise one or more microphones for setting speaker calibrations.

Additional Method Embodiments

[0075] Another example method **1200** for initializing or calibrating a plurality of speakers in a 3-D acoustical reproduction system is shown in FIG. **12**. According to an example embodiment of the invention, the method **1200** starts in block **1202** and includes positioning one or more calibration microphones near a listener position. In block **1204**, the method includes generating calibration tones. In block **1206**, the method includes, selectively routing calibration tones to one or more of the plurality of speakers. The method continues in block **1208** where it includes producing audible tones from the plurality of speakers based on the generated calibration tones. In block **1210**, the method includes sensing audible tones from the plurality of speakers with the one or more calibration microphones. In block **1212**, the method includes determining one or more parameters associated with the plurality of speakers based on sensing the audible tones. In block **1214**, the method includes modifying settings of the 3-D acoustical reproduction system based on the one or more determined parameters. Method **1200** ends after block **1214**.

[0076] An example method **1300** for controlling the apparent location of sounds in a 3-dimensional listening environment is shown in FIG. **13**. According to an example embodi-

ment of the invention, the method **1300** starts in block **1302** and includes receiving one or more audio channels. In block **1304**, the method includes receiving decode data associated with the one or more audio channels. In block **1306**, the method includes routing the one or more audio channels to a plurality of processing channels. In block **1308**, the method includes selectively processing audio associated with the plurality of processing channels based at least in part on the received decode data. In block **1310**, the method includes outputting processed audio to a plurality of speakers. The method **1300** ends after block **1310**.

[0077] Another example method **1400** for recording multi-dimensional audio is shown in FIG. **14**. The method **1400** begins in block **1402** and may include orienting a three-dimensional (3-D) microphone with respect to a predetermined spatial direction. In block **1404**, the method includes selectively receiving sounds from one or more directions corresponding to directional receiving elements. In block **1406**, the method includes recording the selectively received sounds in a 3-D recorder having a plurality of recording channels. In block **1408**, the method includes recording time code in at least one channel of the 3-D recorder. And in block **1410**, the method includes mapping the recorded channels to a plurality of output channels. The method ends after block **1410**.

[0078] The configuration and arrangement of the modules shown and described with respect to the accompanying figures are shown by way of example only, and other configurations and arrangements of system modules can exist in accordance with other embodiments of the invention.

[0079] According to an example embodiment, the invention may be designed specifically for computer gaming and home use. According to another example embodiment, the invention may be designed for professional audio applications, such as in theaters and concert halls.

[0080] Embodiments of the invention can provide various technical effects which may be beneficial for listeners and others. In one aspect of an embodiment of the invention, example systems and methods, when calibrated correctly, may sound about twice as loud (+6 dB) as stereo and/or surround sound yet may only be approximately one sixth (+1 dB) louder.

[0081] In another aspect of an embodiment of the invention, example systems and methods may provide less penetration of walls, floors, and ceilings compared to conventional stereo or surround sound even though they may be approximately one-sixth louder. In this manner, an improved sound system can be provided for apartments, hotels, condos, multiplex theaters, and homes where people outside of the listening environment may want to enjoy relative quiet.

[0082] In another aspect of an embodiment of the invention, example systems and methods can operate with standard conventional sound formats from stereo to surround sound.

[0083] In another aspect of an embodiment of the invention, example systems and methods can operate with a variety of conventional sound sources including, but not limited to, radio, television, cable, satellite radio, digital radio, CDs, DVDs, DVRs, video games, cassettes, records, Blue Ray, etc.

[0084] In another aspect of an embodiment of the invention, example systems and methods may alter the phase to create a sense of 3-D movement.

[0085] The methods disclosed herein are by way of example only, and other methods in accordance with embodiments of the invention can include other elements or steps, including fewer or greater numbers of element or steps than the example methods described herein as well as various combinations of these or other elements.

[0086] While the above description contains many specifics, these specifics should not be construed as limitations on the scope of the invention, but merely as exemplifications of the disclosed embodiments. Those skilled in the art will envision many other possible variations that are within the scope of the invention.

[0087] The invention is described above with reference to block and flow diagrams of systems, methods, apparatuses, and/or computer program products according to example embodiments of the invention. It will be understood that one or more blocks of the block diagrams and flow diagrams, and combinations of blocks in the block diagrams and flow diagrams, respectively, can be implemented by computer-executable program instructions. Likewise, some blocks of the block diagrams and flow diagrams may not necessarily need to be performed in the order presented, or may not necessarily need to be performed at all, according to some embodiments of the invention.

[0088] These computer-executable program instructions may be loaded onto a general purpose computer, a special-purpose computer, a processor, or other programmable data processing apparatus to produce a particular machine, such that the instructions that execute on the computer, processor, or other programmable data processing apparatus create means for implementing one or more functions specified in the flowchart block or blocks. These computer program instructions may also be stored in a computer-readable memory that can direct a computer or other programmable data processing apparatus to function in a particular manner, such that the instructions stored in the computer-readable memory produce an article of manufacture including instruction means that implement one or more functions specified in the flow diagram block or blocks. As an example, embodiments of the invention may provide for a computer program product, comprising a computer usable medium having a computer readable program code or program instructions embodied therein, said computer readable program code adapted to be executed to implement one or more functions specified in the flow diagram block or blocks. The computer program instructions may also be loaded onto a computer or other programmable data processing apparatus to cause a series of operational elements or steps to be performed on the computer or other programmable apparatus to produce a computer-implemented process such that the instructions that execute on the computer or other programmable apparatus provide elements or steps for implementing the functions specified in the flow diagram block or blocks.

[0089] Accordingly, blocks of the block diagrams and flow diagrams support combinations of means for performing the specified functions, combinations of elements or steps for performing the specified functions and program instruction means for performing the specified functions. It will also be understood that each block of the block diagrams and flow diagrams, and combinations of blocks in the block diagrams and flow diagrams, can be implemented by special-purpose, hardware-based computer systems that perform the specified functions, elements or steps, or combinations of special purpose hardware and computer instructions.

[0090] In certain embodiments, performing the specified functions, elements or steps can transform an article into another state or thing. For instance, example embodiments of

the invention can provide certain systems and methods that transform encoded audio electronic signals into time-varying sound pressure levels. Example embodiments of the invention can provide the further systems and methods for that transform positional information to directional audio.

[0091] Many modifications and other embodiments of the invention set forth herein will be apparent having the benefit of the teachings presented in the foregoing descriptions and the associated drawings. Therefore, it is to be understood that the invention is not to be limited to the specific embodiments disclosed and that modifications and other embodiments are intended to be included within the scope of the appended claims. Although specific terms are employed herein, they are used in a generic and descriptive sense only and not for purposes of limitation.

The claimed invention is:

1. A method for initializing a plurality of speakers in a 3D acoustical reproduction system, the method comprising:
- positioning one or more calibration microphones near a listener position;
- generating calibration tones;
- selectively routing calibration tones to one or more of the plurality of speakers;
- producing audible tones from the plurality of speakers based on the generated calibration tones;
- sensing audible tones from the plurality of speakers with the one or more calibration microphones;
- determining one or more parameters associated with the plurality of speakers based on sensing the audible tones; and
- modifying settings of the 3D acoustical reproduction system based on the one or more determined parameters.

2. The method of claim **1**, wherein positioning comprises placing the one or more calibration microphones within about 10 cm of a listener position.

3. The method of claim **1**, wherein generating calibration tones comprises generating one or more of impulses, chirps, white noise, pink noise, tone warbling, modulated tones, phase shifted tones, multiple tones or audible prompts.

4. The method of claim **1**, wherein selectively routing calibration tones comprises routing signals individually or in combination to the plurality of speakers.

5. The method of claim **1**, wherein producing audible tones from the plurality of speakers comprises amplifying the calibration tones.

6. The method of claim **1**, wherein the one or more parameters associated with the plurality of speakers comprise one or more of phase, delay, frequency response, impulse response, distance from the one or more calibration microphones, position with respect to the one or more calibration microphones, speaker axial angle, speaker radial angle, or speaker azimuth angle.

7. The method of claim **1**, wherein modifying settings comprise modifying one or more of volume, equalization, or delay.

8. The method of claim **1**, wherein modifying settings comprises storing calibration parameters in memory.

9. A system for initializing a plurality of speakers, the system comprising:
- a plurality of speakers;
- at least one calibration tone generator;
- one or more calibration microphones for sensing audible tones from the plurality of speakers;
- at least one memory; and
- one or more processors in communication with the memory and configured for
  - selectively generating and routing calibration tones to one or more of the plurality of speakers for producing audible tones;
  - determining one or more parameters associated with the plurality of speakers based on sensing the audible tones; and
  - modifying settings of the 3D acoustical reproduction system based on the one or more determined parameters.

10. The system of claim **9**, wherein the one or more calibration microphones are wireless.

11. The system of claim **9**, wherein the one or more calibration microphones are positioned within about 10 cm of a listener position.

12. The system of claim **9**, wherein the one or more processors are further configured to selectively generate calibration tones, wherein the calibration tones comprise one or more of impulses, chirps, white noise, pink noise, tone warbling, modulated tones, phase shifted tones, multiple tones or audible prompts.

13. The system of claim **9**, wherein the one or more processors are further configured to selectively route calibration tones individually or in combination to the plurality of speakers.

14. The system of claim **9**, further comprising a plurality of amplifiers for amplifying the calibration tones.

15. The system of claim **9**, wherein the one or more parameters associated with the plurality of speakers comprise one or more of phase, delay, frequency response, impulse response, distance from the one or more calibration microphones, position with respect to the one or more calibration microphones, speaker axial angle, speaker radial angle, or speaker azimuth angle.

16. The system of claim **9**, wherein modifying settings comprises modifying one or more of volume, equalization, or delay.

17. The system of claim **9**, wherein modifying settings comprises storing calibration parameters the memory.

18. An apparatus for providing multi-dimensional audio, the apparatus comprising:
- one or more input interfaces for receiving signals associated with multi-dimensional audio;
- one or more output interfaces; and
- a processor in communication with the one or more input interfaces and the one or more output interfaces, the processor comprising:
  - at least one analog to digital converter;
  - a signal router; and
  - six or more programmable preamps.

19. The apparatus of claim **18**, further comprising six or more output amplifiers in communication with the processor and the one or more output interfaces.

20. The apparatus of claim **19**, further comprising at lest one digital to analog converter.

* * * * *