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REPUBLICATION

(54) **COMPUTER, INTERNET AND TELECOMMUNICATIONS BASED NETWORK**

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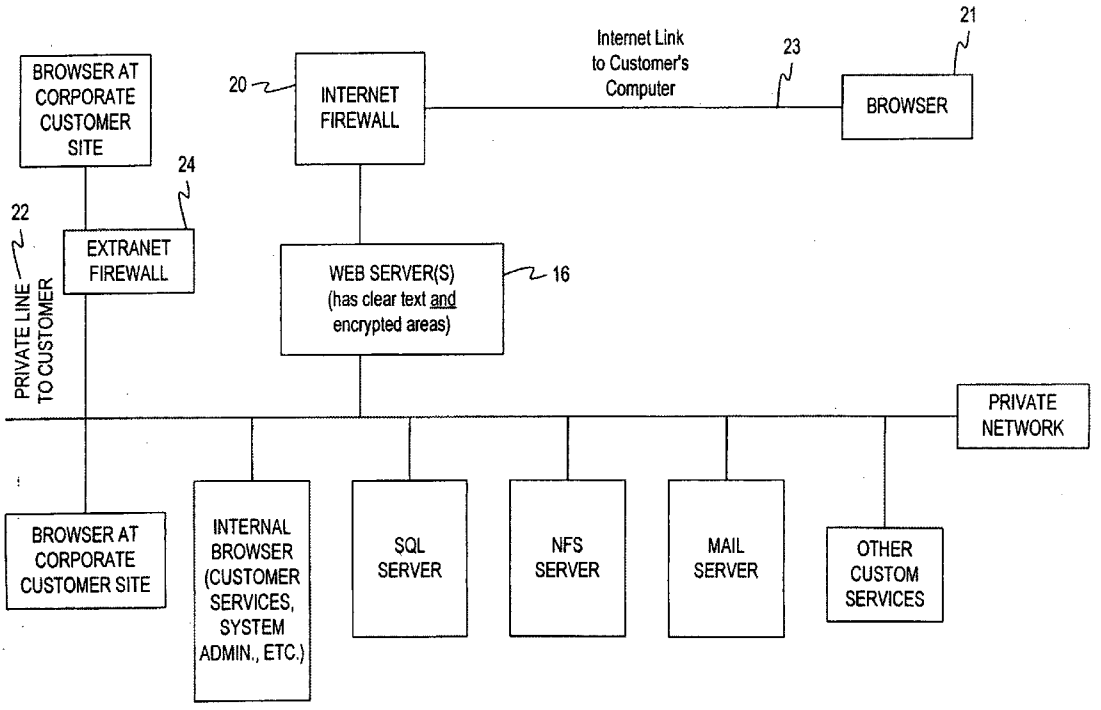
(57) **ABSTRACT**

A method and apparatus for a computer and telecommunication network which can receive, send and manage information from or to a subscriber of the network, based on the subscriber's configuration. The network is made up of at least one cluster containing voice servers which allow for telephony, speech recognition, text-to-speech and conferencing functions, and is accessible by the subscriber through standard telephone connections or through internet connections. The network also utilizes a database and file server allowing the subscriber to maintain and manage certain contact lists and administrative information. A web server is also connected to the cluster, thereby allowing access to all functions through internet connections.



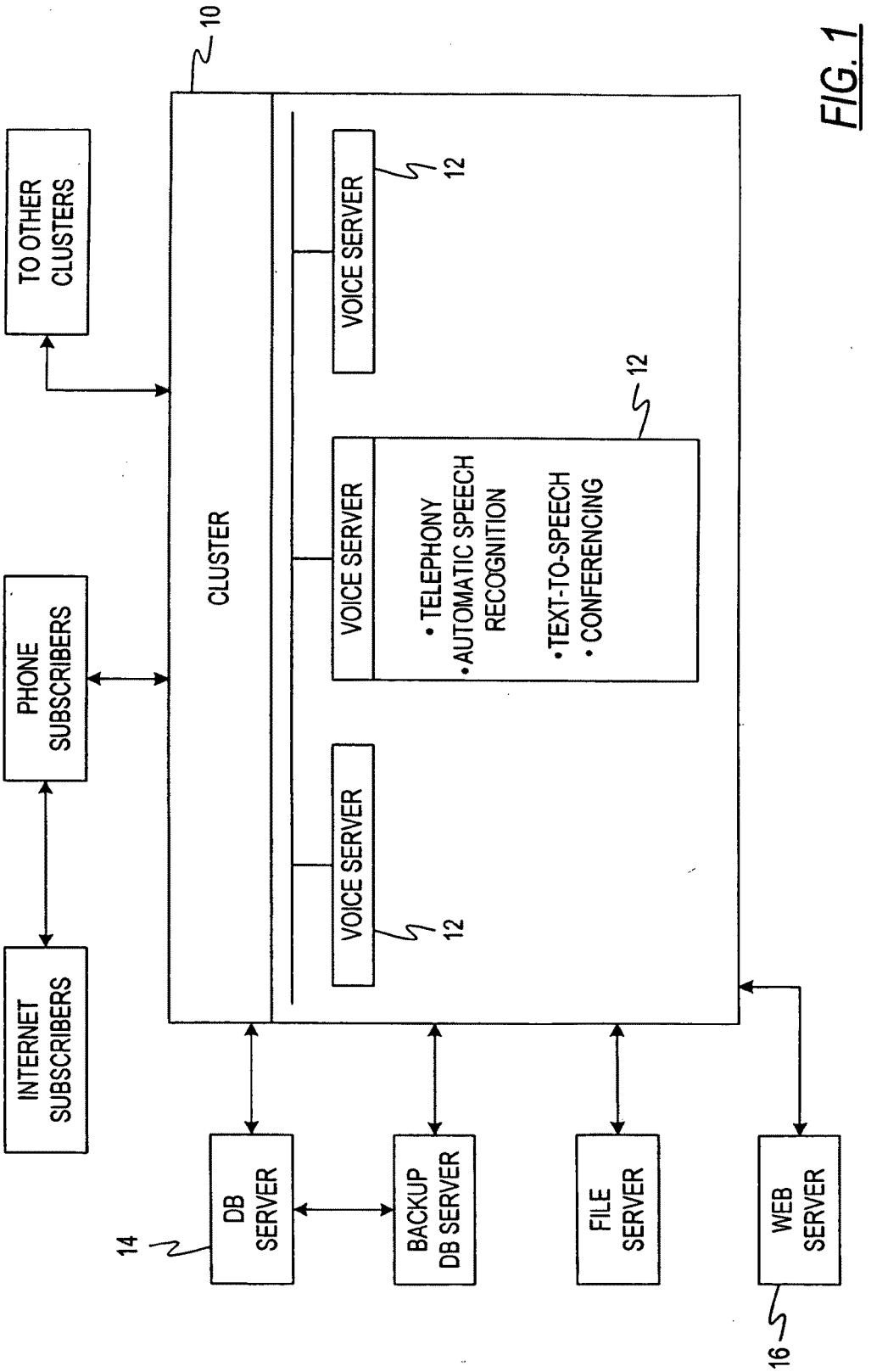

10
TO OTHER CLUSTERS
PHONE SUBSCRIBERS
INTERNET SUBSCRIBERS
CLUSTER
VOICE SERVER
12
VOICE SERVER
12
VOICE SERVER
12
• TELEPHONY
• AUTOMATIC SPEECH RECOGNITION
• TEXT-TO-SPEECH
• CONFERENCING
14
DB SERVER
BACKUP DB SERVER
FILE SERVER
WEB SERVER
16

FIG. 1

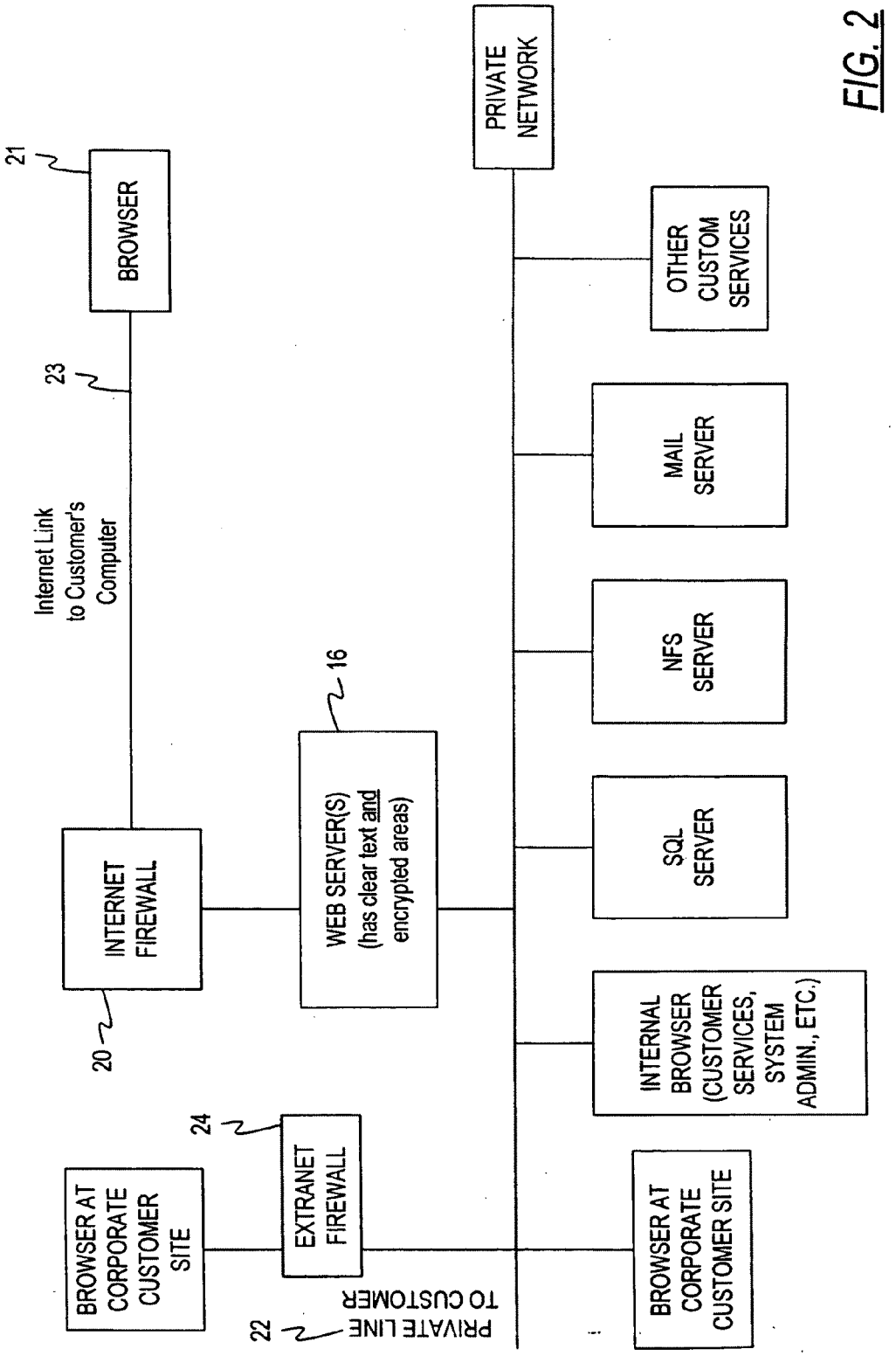

PRIVATE NETWORK
21
BROWSER
23
Internet Link to Customer's Computer
INTERNET FIREWALL
20
16
WEB SERVER(S) (has clear text and encrypted areas)
24
EXTRANET FIREWALL
22
PRIVATE LINE TO CUSTOMER
BROWSER AT CORPORATE CUSTOMER SITE
BROWSER AT CORPORATE CUSTOMER SITE
INTERNAL BROWSER (CUSTOMER SERVICES, SYSTEM ADMIN., ETC.)
SQL SERVER
NFS SERVER
MAIL SERVER
OTHER CUSTOM SERVICES

FIG. 2

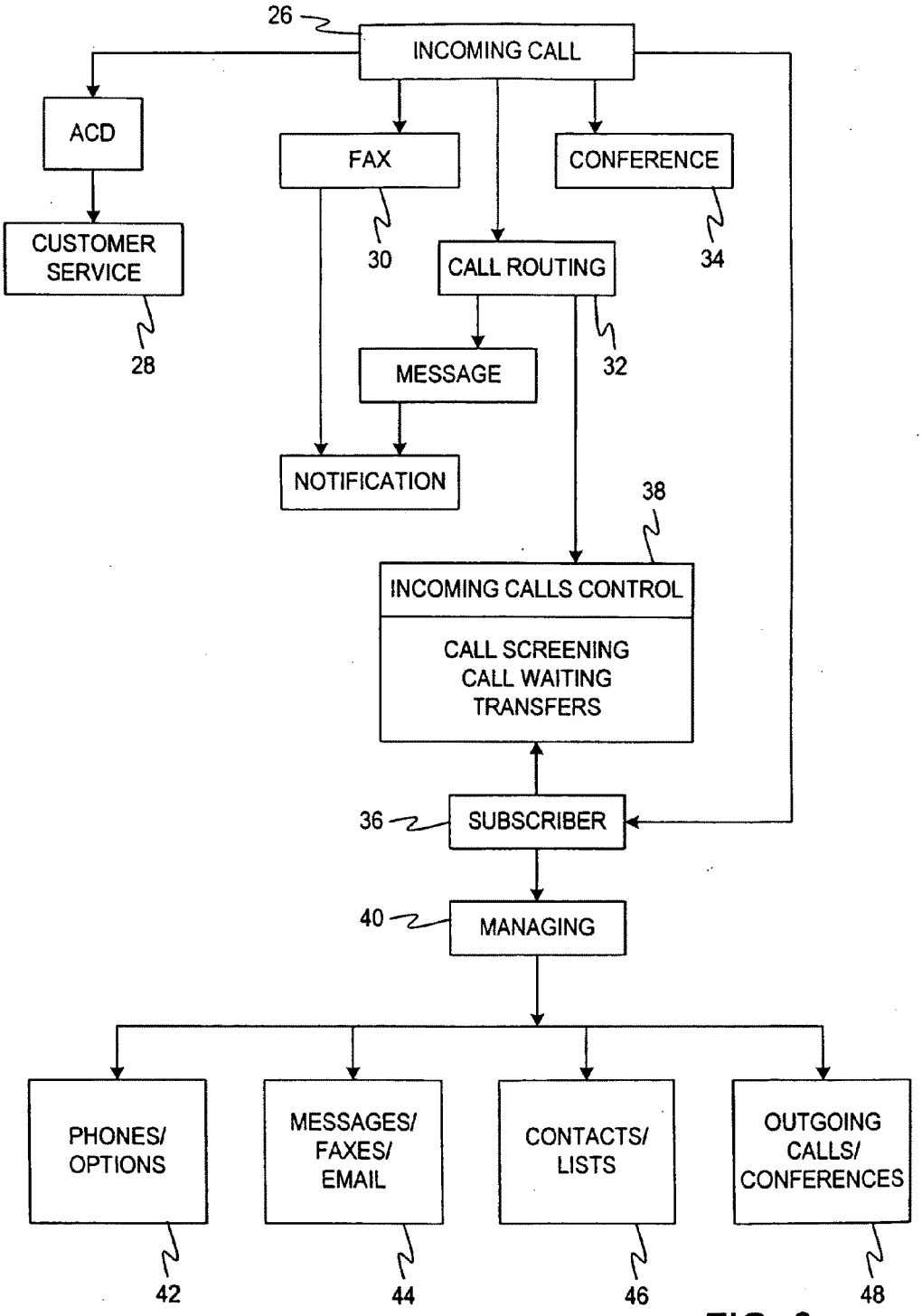

26
INCOMING CALL
ACD
FAX
CONFERENCE
CUSTOMER SERVICE
30
CALL ROUTING
34
28
MESSAGE
32
NOTIFICATION
38
INCOMING CALLS CONTROL
CALL SCREENING
CALL WAITING
TRANSFERS
36
SUBSCRIBER
40
MANAGING
PHONES/ OPTIONS
MESSAGES/ FAXES/ EMAIL
CONTACTS/ LISTS
OUTGOING CALLS/ CONFERENCES
42
44
46
48

FIG. 3

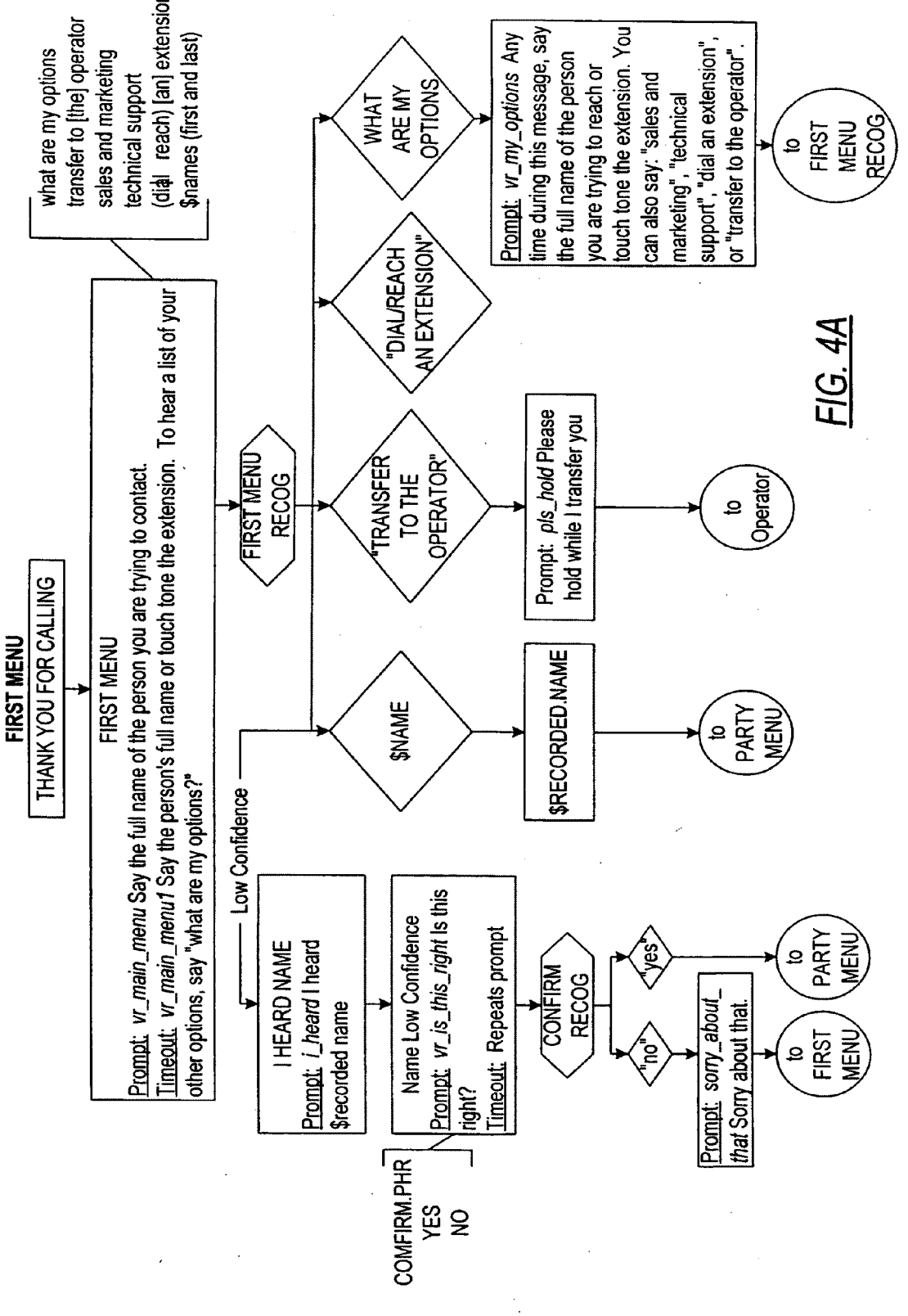

FIRST MENU
THANK YOU FOR CALLING
FIRST MENU
Prompt: vr_main_menu Say the full name of the person you are trying to contact.
Timeout: vr_main_menu1 Say the person's full name or touch tone the extension. To hear a list of your other options, say "what are my options?"
what are my options
transfer to [the] operator
sales and marketing
technical support
(dial reach) [an] extension
$names (first and last)
FIRST MENU RECOG
Low Confidence
I HEARD NAME
Prompt: i_heard I heard $recorded name
Name Low Confidence
Prompt: vr_is_this_right Is this right?
Timeout: Repeats prompt
COMFIRM.PHR
YES
NO
CONFIRM RECOG
"yes"
"no"
Prompt: sorry_about_that Sorry about that.
to FIRST MENU
to PARTY MENU
$NAME
$RECORDED.NAME
to PARTY MENU
"TRANSFER TO THE OPERATOR"
Prompt: pls_hold Please hold while I transfer you
to Operator
"DIAL/REACH AN EXTENSION"
WHAT ARE MY OPTIONS
Prompt: vr_my_options Any time during this message, say the full name of the person you are trying to reach or touch tone the extension. You can also say: "sales and marketing", "technical support", "dial an extension", or "transfer to the operator".
to FIRST MENU RECOG

FIG. 4A

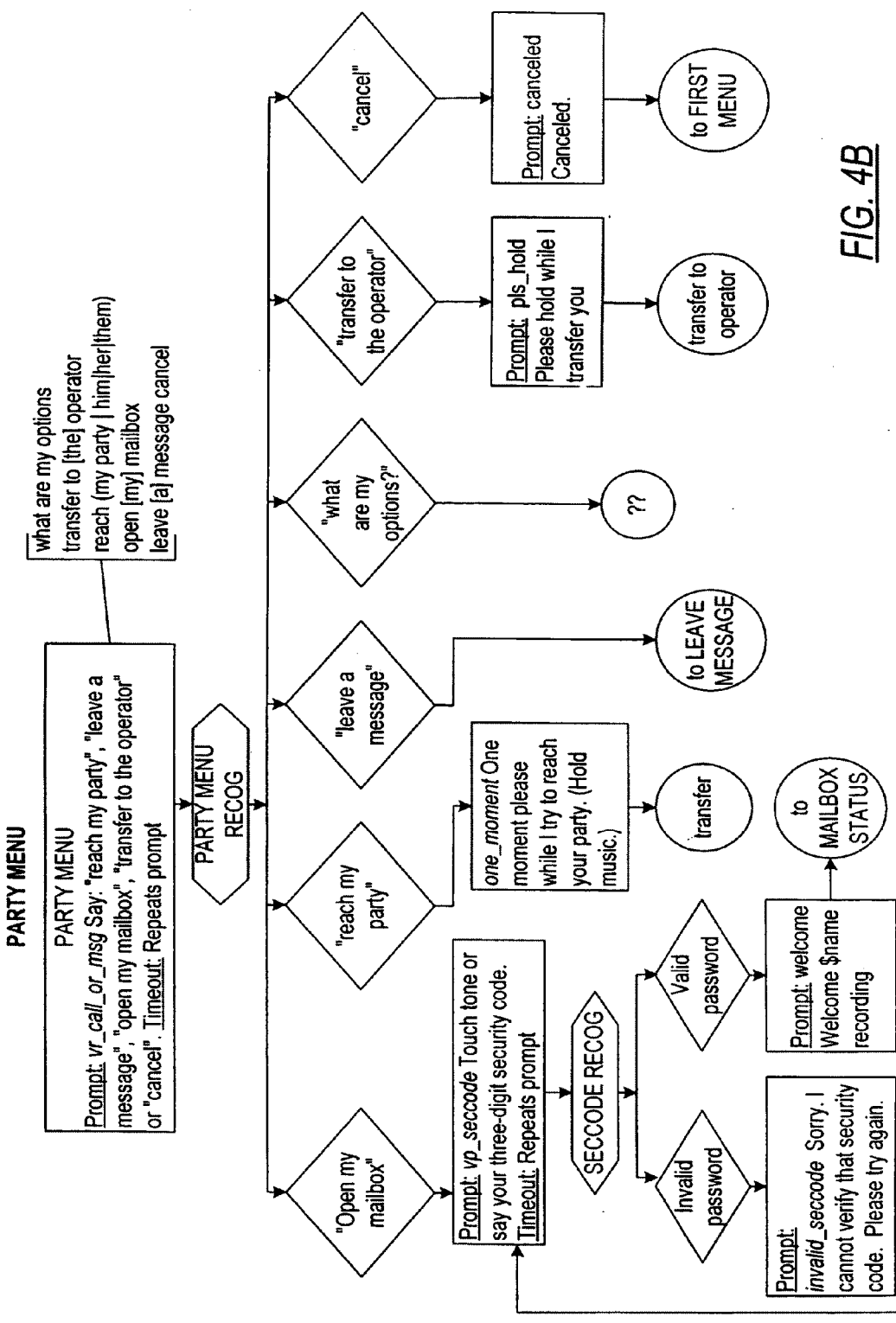

PARTY MENU
PARTY MENU
Prompt: vr_call_or_msg Say: "reach my party", "leave a message", "open my mailbox", "transfer to the operator" or "cancel". Timeout: Repeats prompt
what are my options
transfer to [the] operator
reach (my party | him|her|them)
open [my] mailbox
leave [a] message cancel
PARTY MENU RECOG
"Open my mailbox"
"reach my party"
"leave a message"
"what are my options?"
"transfer to the operator"
"cancel"
Prompt: vp_seccode Touch tone or say your three-digit security code. Timeout: Repeats prompt
SECCODE RECOG
Invalid password
Valid password
Prompt: invalid_seccode Sorry. I cannot verify that security code. Please try again.
Prompt: welcome Welcome $name recording
to MAILBOX STATUS
one_moment One moment please while I try to reach your party. (Hold music.)
transfer
to LEAVE MESSAGE
??
Prompt: pls_hold Please hold while I transfer you
transfer to operator
Prompt: canceled Canceled.
to FIRST MENU

FIG. 4B

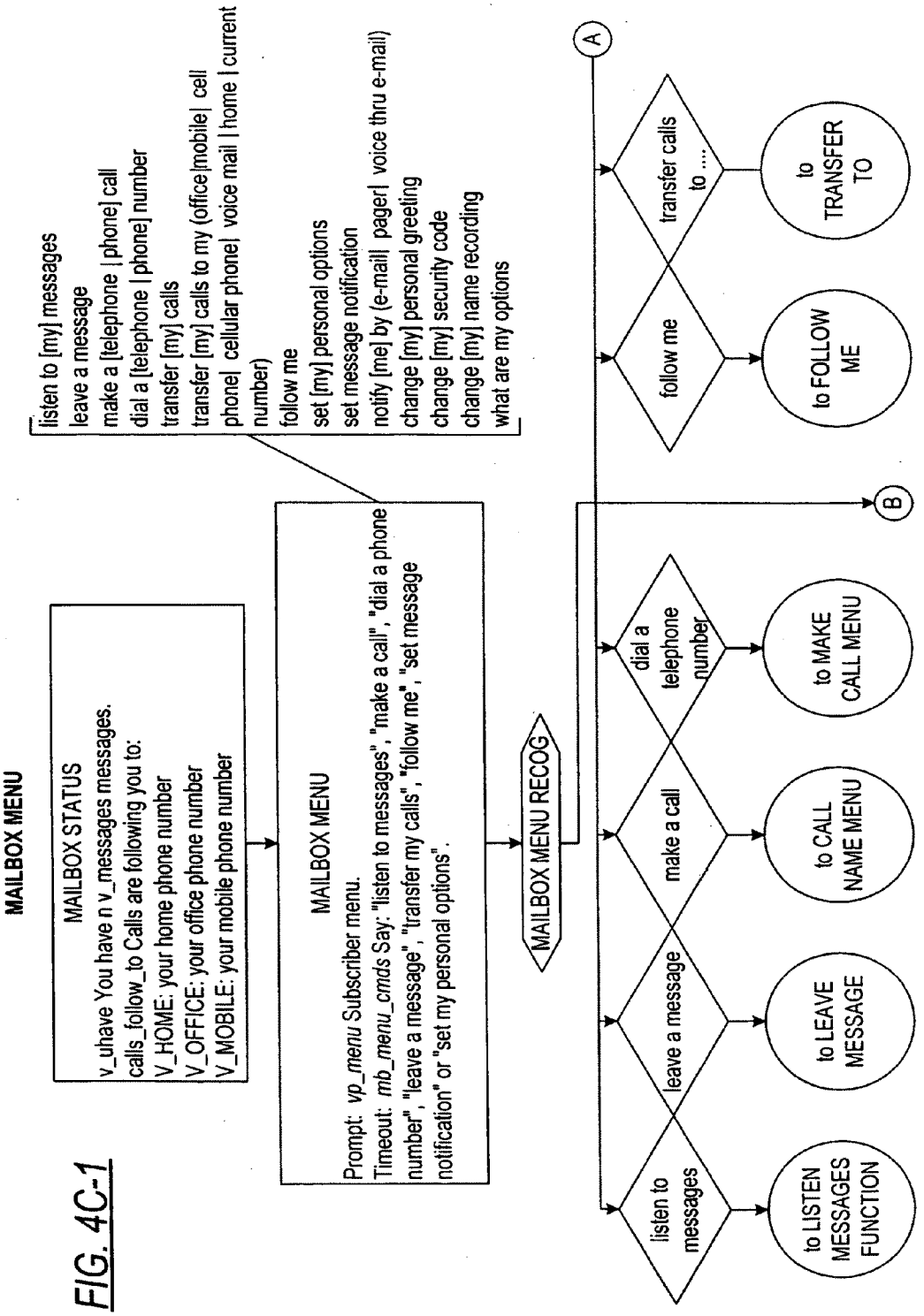

MAILBOX MENU
MAILBOX STATUS
v_uhave You have n v_messages messages.
calls_follow_to Calls are following you to:
V_HOME: your home phone number
V_OFFICE: your office phone number
V_MOBILE: your mobile phone number
MAILBOX MENU
Prompt: vp_menu Subscriber menu.
Timeout: mb_menu_cmds Say: "listen to messages", "make a call", "dial a phone number", "leave a message", "transfer my calls", "follow me", "set message notification" or "set my personal options".
listen to [my] messages
leave a message
make a [telephone | phone] call
dial a [telephone | phone] number
transfer [my] calls
transfer [my] calls to my (office|mobile| cell phone| cellular phone| voice mail | home | current number)
follow me
set [my] personal options
set message notification
notify [me] by (e-mail| pager| voice thru e-mail)
change [my] personal greeting
change [my] security code
change [my] name recording
what are my options
MAILBOX MENU RECOG
listen to messages
leave a message
make a call
dial a telephone number
follow me
transfer calls to ....
to LISTEN MESSAGES FUNCTION
to LEAVE MESSAGE
to CALL NAME MENU
to MAKE CALL MENU
to FOLLOW ME
to TRANSFER TO
A
B

FIG. 4C-1

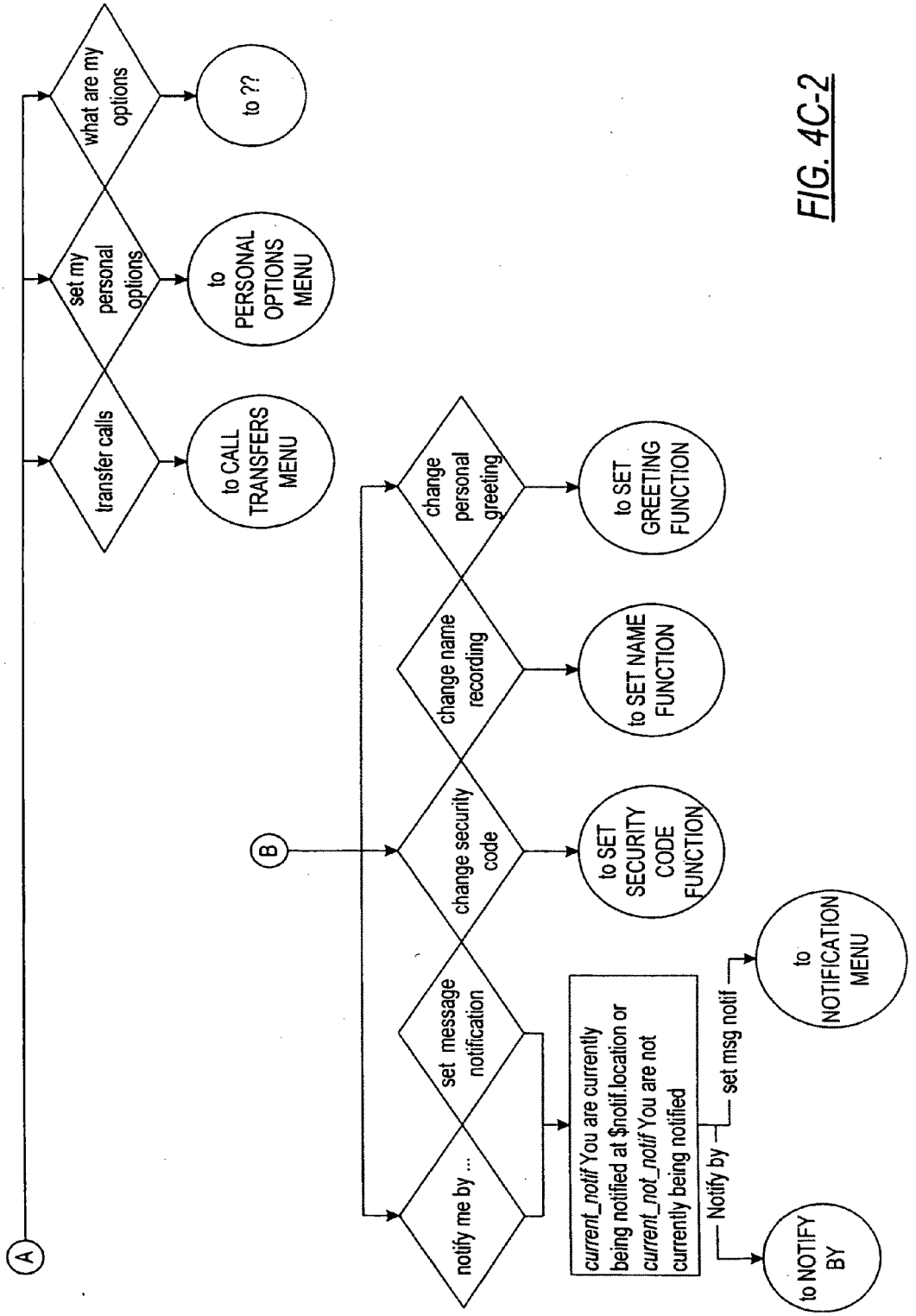

A
what are my options
to ??
set my personal options
to PERSONAL OPTIONS MENU
transfer calls
to CALL TRANSFERS MENU
B
change personal greeting
to SET GREETING FUNCTION
change name recording
to SET NAME FUNCTION
change security code
to SET SECURITY CODE FUNCTION
set message notification
notify me by ...
current_notif You are currently being notified at $notif.location or current_not_notif You are not currently being notified
Notify by
set msg notif
to NOTIFY BY
to NOTIFICATION MENU

FIG. 4C-2

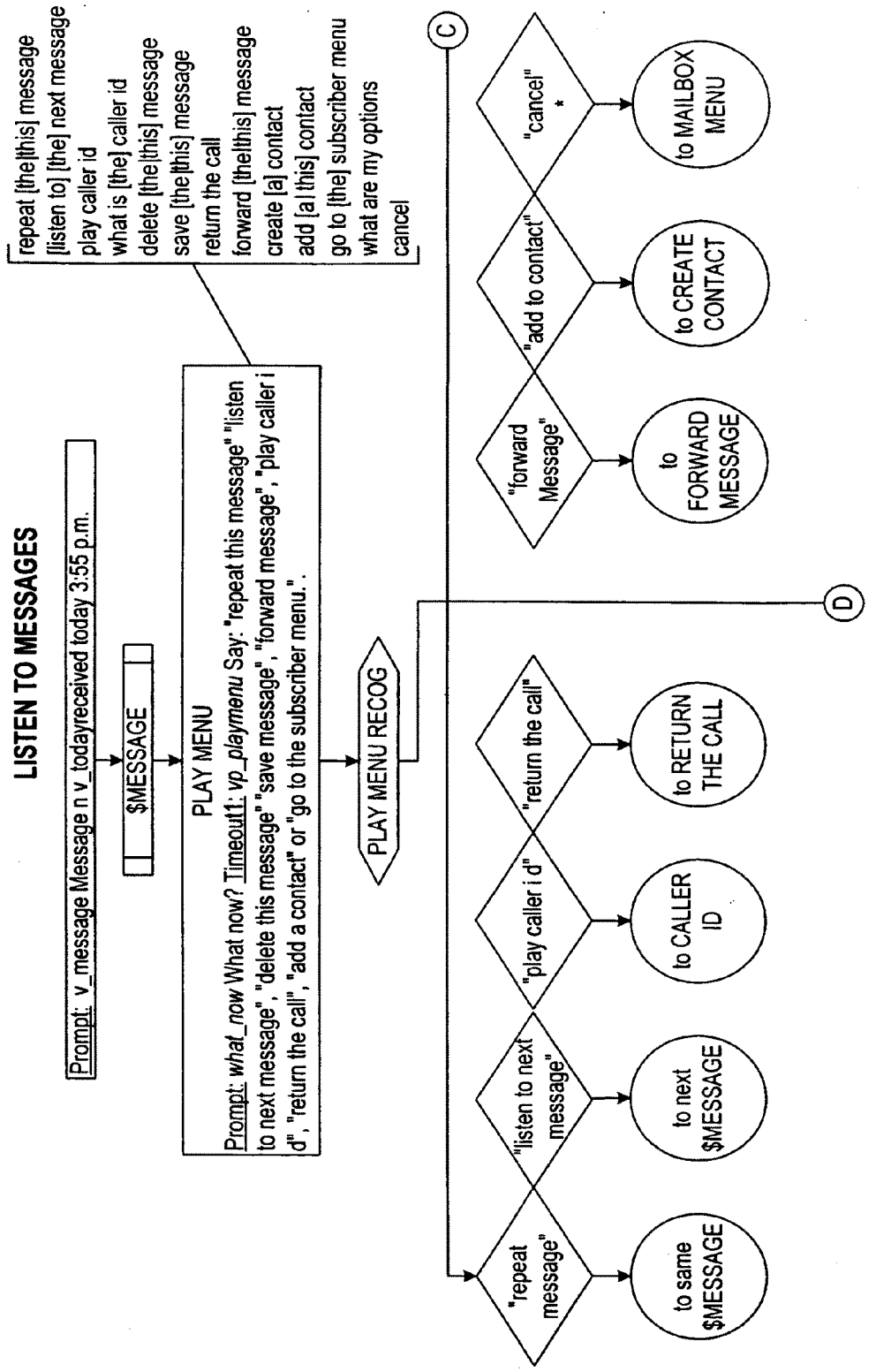
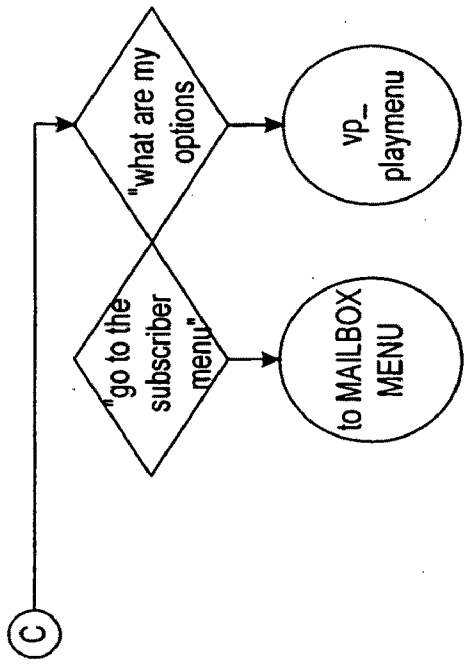

LISTEN TO MESSAGES
Prompt: v_message Message n v_todayreceived today 3:55 p.m.
$MESSAGE
PLAY MENU
Prompt: what_now What now? Timeout1: vp_playmenu Say: "repeat this message" "listen to next message", "delete this message" "save message", "forward message", "play caller i d", "return the call", "add a contact" or "go to the subscriber menu.."
repeat [the|this] message
[listen to] [the] next message
play caller id
what is [the] caller id
delete [the|this] message
save [the|this] message
return the call
forward [the|this] message
create [a] contact
add [a|this] contact
go to [the] subscriber menu
what are my options
cancel
PLAY MENU RECOG
C
D
"repeat message"
to same $MESSAGE
"listen to next message"
to next $MESSAGE
"play caller i d"
to CALLER ID
"return the call"
to RETURN THE CALL
"forward Message"
to FORWARD MESSAGE
"add to contact"
to CREATE CONTACT
"cancel" *
to MAILBOX MENU

FIG. 4D-1


C
"what are my options
vp_ playmenu
"go to the subscriber menu"
to MAILBOX MENU

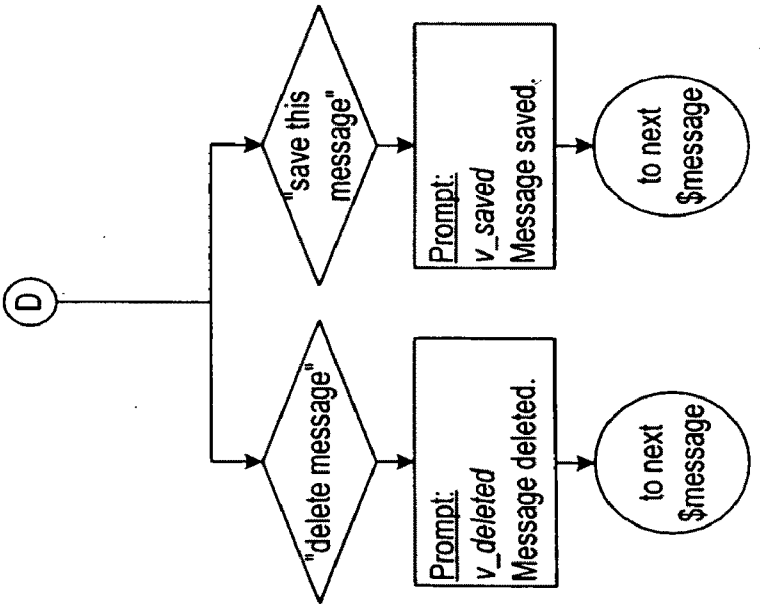

D
"save this message"
Prompt: v_saved Message saved.
to next $message
"delete message"
Prompt: v_deleted Message deleted.
to next $message

FIG. 4D-2

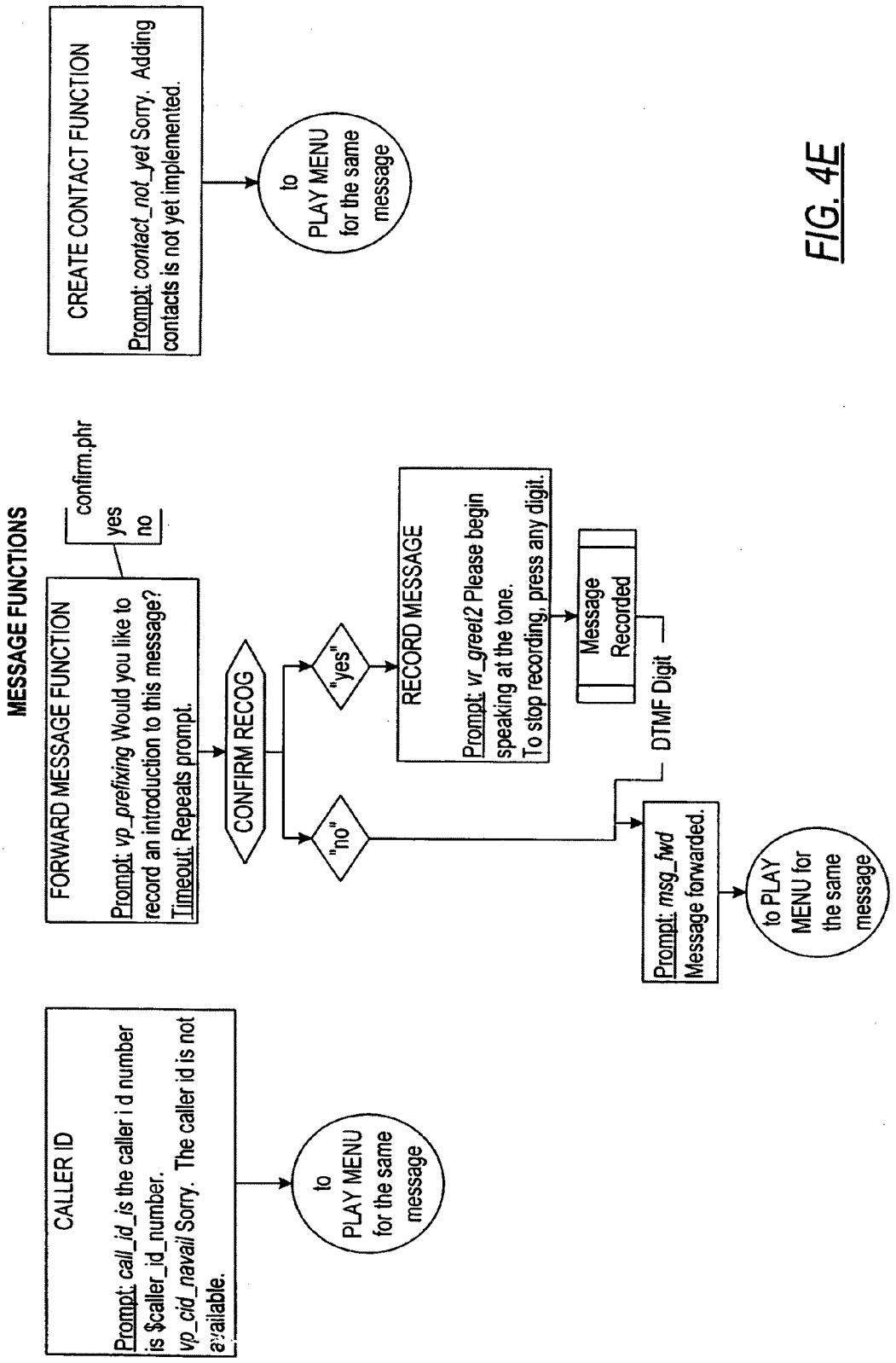

MESSAGE FUNCTIONS
CALLER ID
Prompt: call_id_is the caller i d number is $caller_id_number. vp_cid_navail Sorry. The caller id is not available.
to PLAY MENU for the same message
FORWARD MESSAGE FUNCTION
Prompt: vp_prefixing Would you like to record an introduction to this message?
Timeout: Repeats prompt.
confirm.phr
yes
no
CONFIRM RECOG
"yes"
"no"
RECORD MESSAGE
Prompt: vr_greet2 Please begin speaking at the tone. To stop recording, press any digit.
Message Recorded
DTMF Digit
Prompt: msg_fwd Message forwarded.
to PLAY MENU for the same message
CREATE CONTACT FUNCTION
Prompt: contact_not_yet Sorry. Adding contacts is not yet implemented.
to PLAY MENU for the same message

FIG. 4E

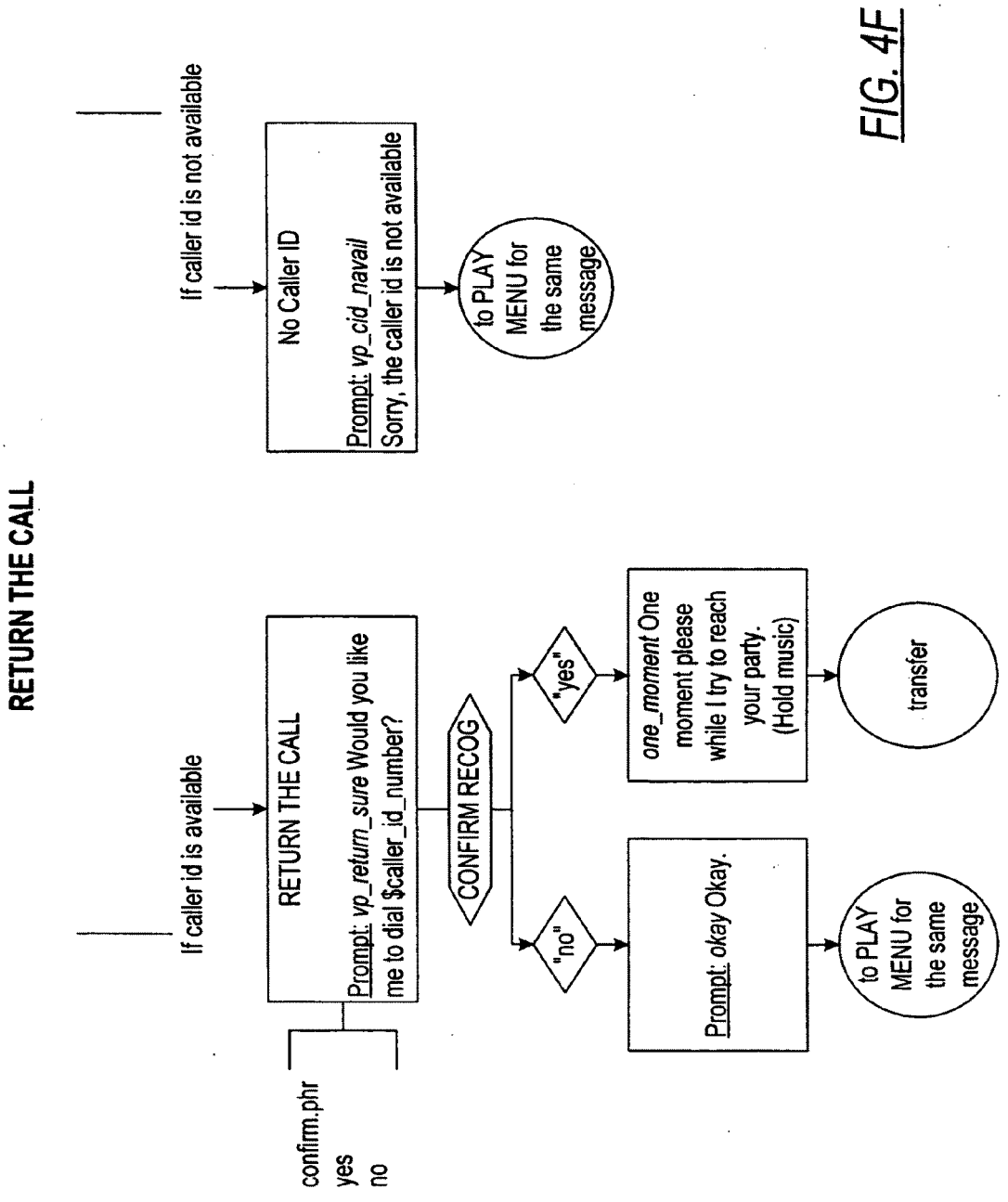

RETURN THE CALL
If caller id is available
RETURN THE CALL
Prompt: vp_return_sure Would you like me to dial $caller_id_number?
confirm.phr
yes
no
CONFIRM RECOG
"yes"
one_moment One moment please while I try to reach your party. (Hold music)
transfer
"no"
Prompt: okay Okay.
to PLAY MENU for the same message
If caller id is not available
No Caller ID
Prompt: vp_cid_navail Sorry, the caller id is not available
to PLAY MENU for the same message

FIG. 4F

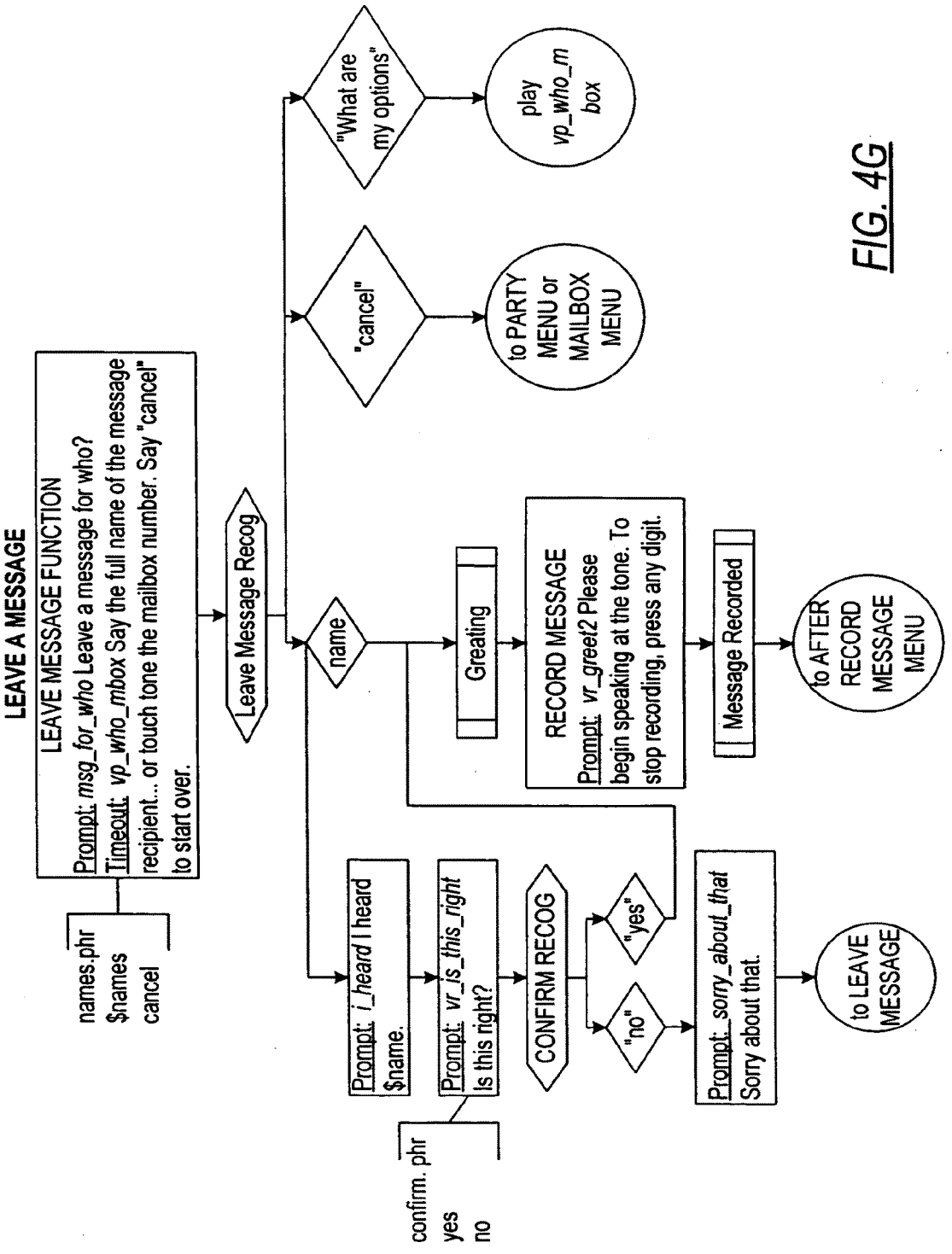

LEAVE A MESSAGE
LEAVE MESSAGE FUNCTION
Prompt: msg_for_who Leave a message for who?
Timeout: vp_who_mbox Say the full name of the message recipient... or touch tone the mailbox number. Say "cancel" to start over.
names.phr
$names
cancel
Leave Message Recog
"What are my options"
play vp_who_m box
"cancel"
to PARTY MENU or MAILBOX MENU
name
Greating
RECORD MESSAGE
Prompt: vr_greet2 Please begin speaking at the tone. To stop recording, press any digit.
Message Recorded
to AFTER RECORD MESSAGE MENU
Prompt: i_heard I heard $name.
Prompt: vr_is_this_right Is this right?
CONFIRM RECOG
"yes"
"no"
Prompt: sorry_about_that Sorry about that.
to LEAVE MESSAGE
confirm. phr
yes
no

FIG. 4G

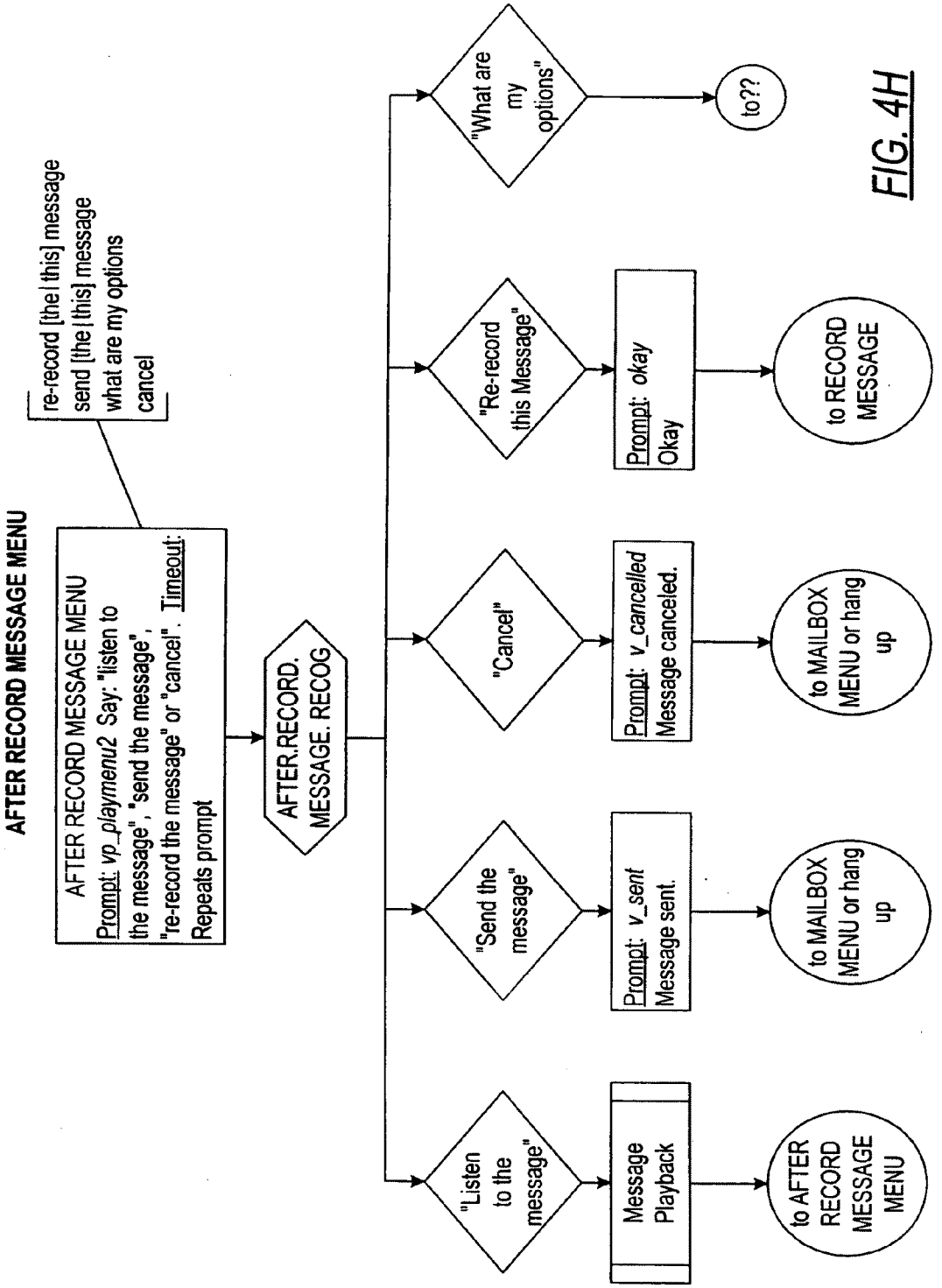

AFTER RECORD MESSAGE MENU
AFTER RECORD MESSAGE MENU
Prompt: vp_playmenu2 Say: "listen to the message", "send the message", "re-record the message" or "cancel". Timeout: Repeats prompt
re-record [the | this] message
send [the | this] message
what are my options
cancel
AFTER.RECORD. MESSAGE. RECOG
"Listen to the message"
Message Playback
to AFTER RECORD MESSAGE MENU
"Send the message"
Prompt: v_sent Message sent.
to MAILBOX MENU or hang up
"Cancel"
Prompt: v_cancelled Message canceled.
to MAILBOX MENU or hang up
"Re-record this Message"
Prompt: okay Okay
to RECORD MESSAGE
"What are my options"
to??

FIG. 4H

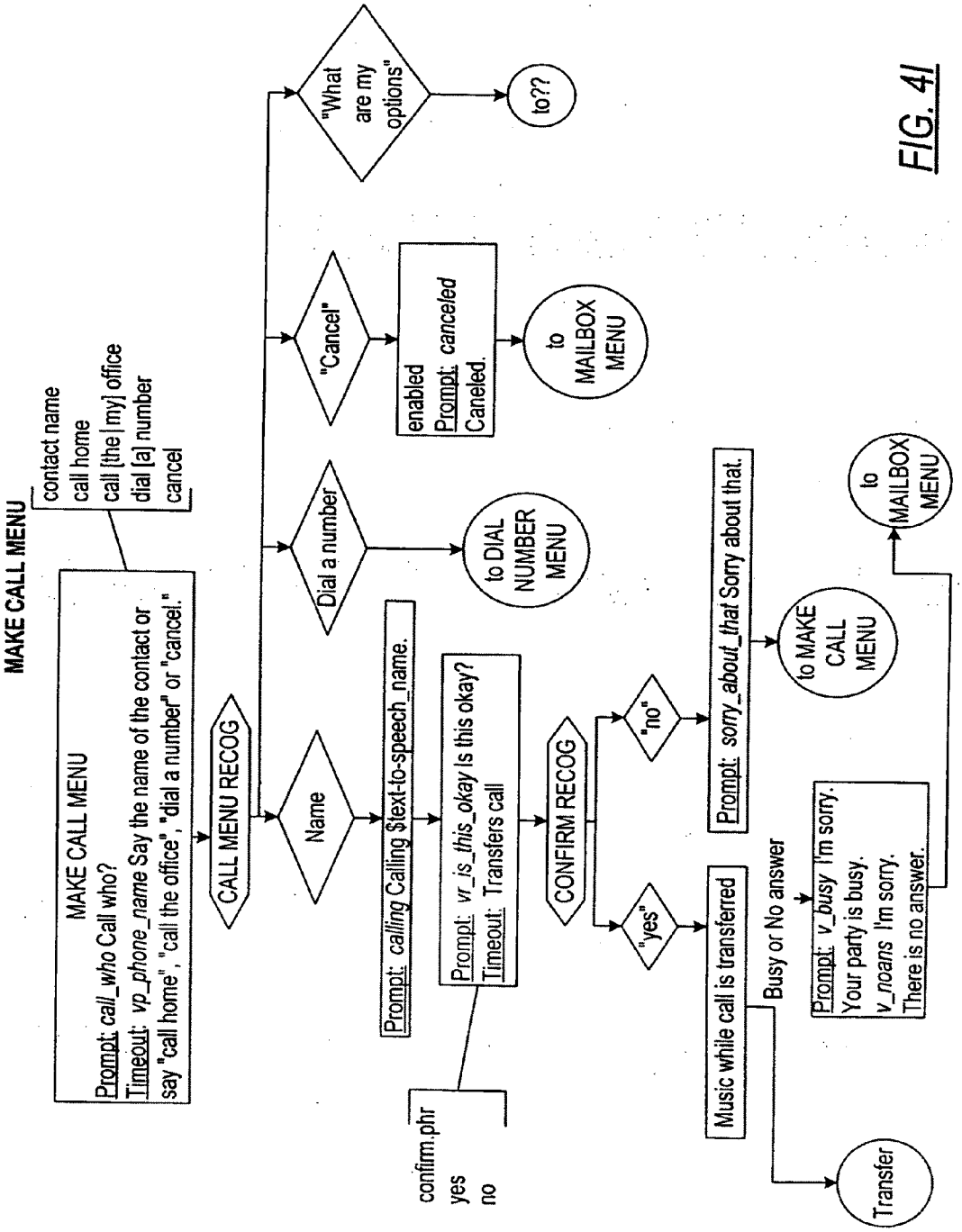

MAKE CALL MENU
MAKE CALL MENU
Prompt: call_who Call who?
Timeout: vp_phone_name Say the name of the contact or say "call home", "call the office", "dial a number" or "cancel."
contact name
call home
call [the | my] office
dial [a] number
cancel
CALL MENU RECOG
Name
Prompt: calling Calling $text-to-speech_name.
Prompt: vr_is_this_okay Is this okay?
Timeout: Transfers call
confirm.phr
yes
no
CONFIRM RECOG
"yes"
"no"
Music while call is transferred
Busy or No answer
Prompt: v_busy I'm sorry. Your party is busy. v_noans I'm sorry. There is no answer.
Prompt: sorry_about_that Sorry about that.
to MAKE CALL MENU
to MAILBOX MENU
Transfer
Dial a number
to DIAL NUMBER MENU
"Cancel"
enabled
Prompt: canceled
Canceled.
to MAILBOX MENU
"What are my options"
to??

FIG. 4I

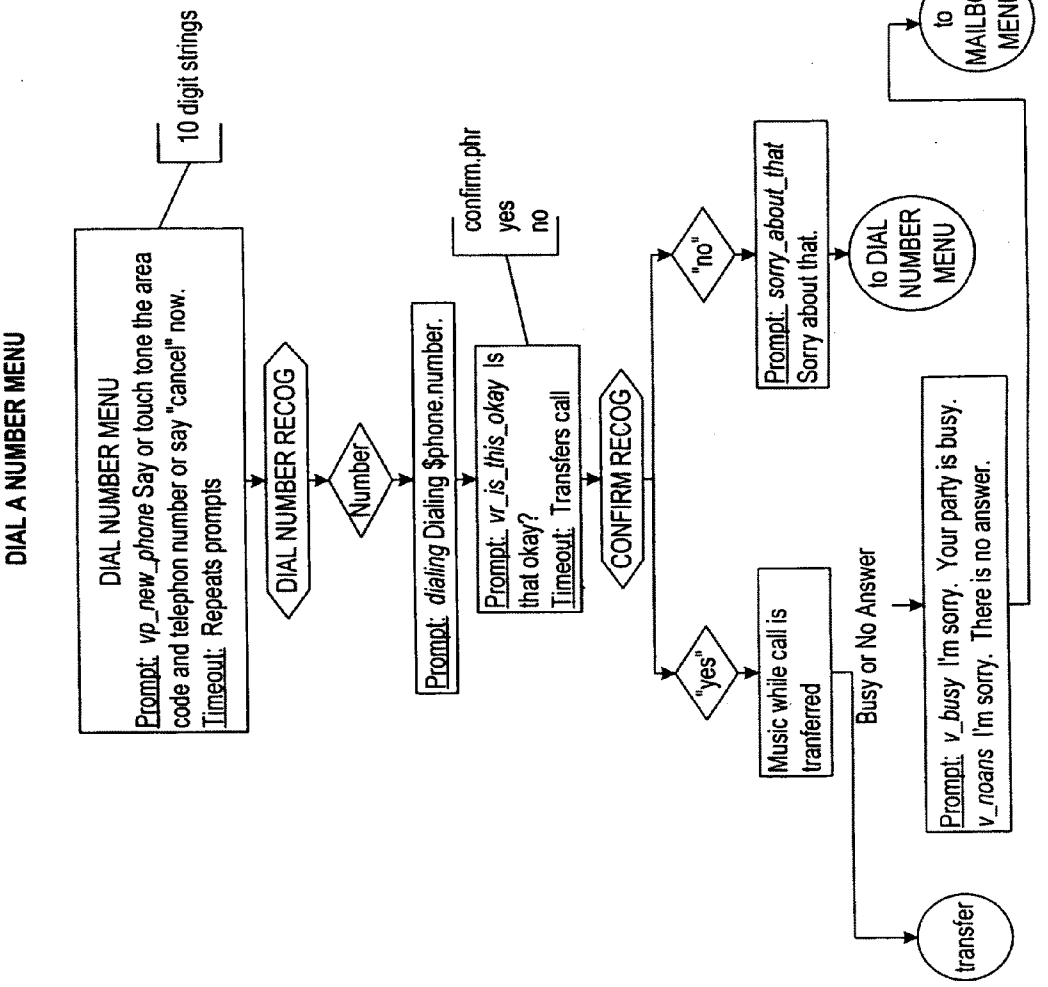

DIAL A NUMBER MENU
DIAL NUMBER MENU
Prompt: vp_new_phone Say or touch tone the area code and telephon number or say "cancel" now.
Timeout: Repeats prompts
10 digit strings
DIAL NUMBER RECOG
Number
Prompt: dialing Dialing $phone.number.
Prompt: vr_is_this_okay Is that okay?
Timeout: Transfers call
confirm.phr
yes
no
CONFIRM RECOG
"yes"
"no"
Music while call is tranferred
Prompt: sorry_about_that Sorry about that.
to DIAL NUMBER MENU
to MAILBOX MENU
Busy or No Answer
Prompt: v_busy I'm sorry. Your party is busy.
v_noans I'm sorry. There is no answer.
transfer

FIG. 4J

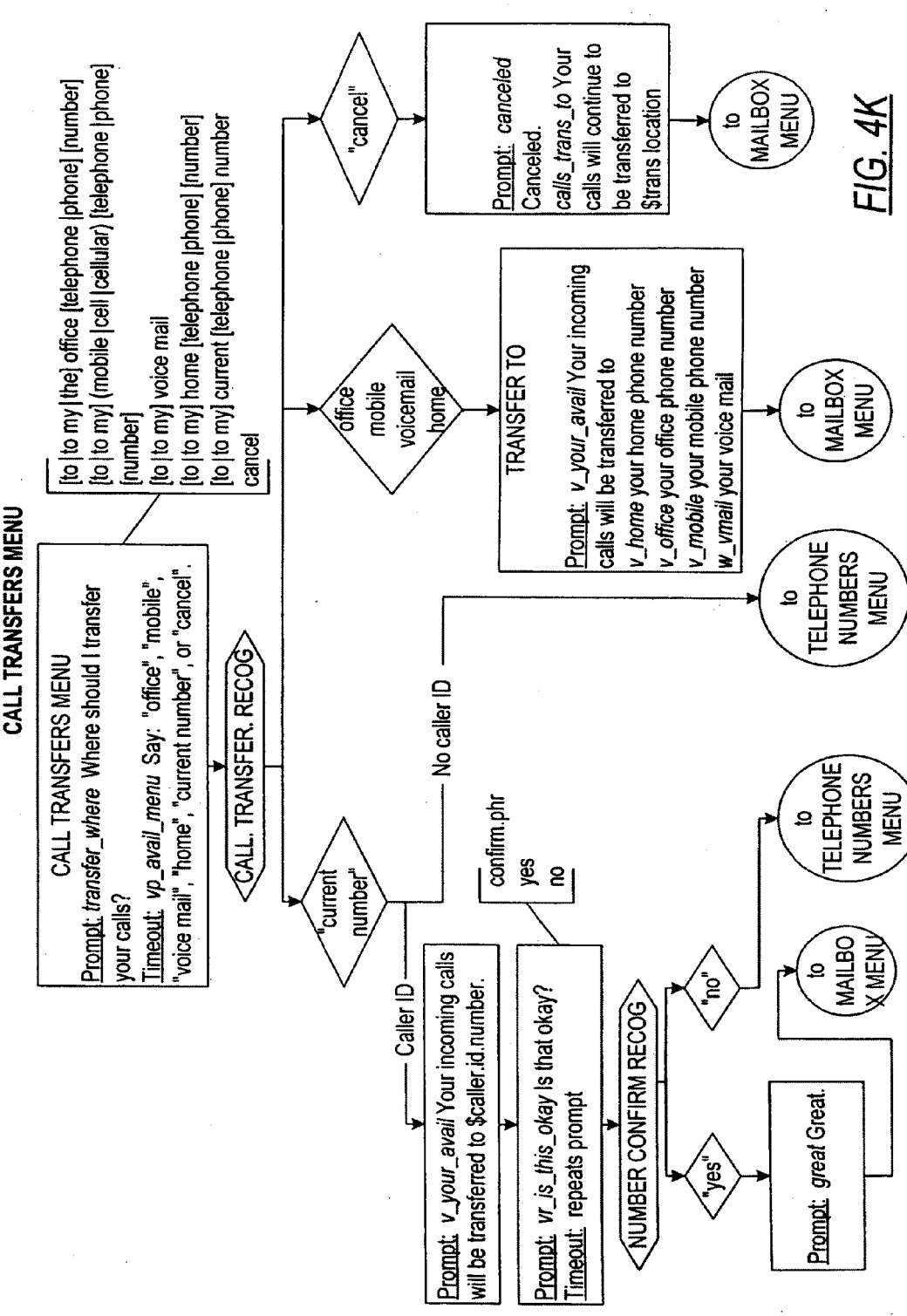

CALL TRANSFERS MENU
CALL TRANSFERS MENU
Prompt: transfer_where Where should I transfer your calls?
Timeout: vp_avail_menu Say: "office", "mobile", "voice mail", "home", "current number", or "cancel".
[to | to my | the] office [telephone |phone] [number]
[to | to my] (mobile |cell |cellular) [telephone |phone] [number]
[to | to my] voice mail
[to | to my] home [telephone |phone] [number]
[to | to my] current [telephone |phone] number
cancel
CALL. TRANSFER. RECOG
"cancel"
Prompt: canceled
Canceled.
calls_trans_to Your calls will continue to be transferred to $trans location
to MAILBOX MENU
office
mobile
voicemail
home
TRANSFER TO
Prompt: v_your_avail Your incoming calls will be transferred to
v_home your home phone number
v_office your office phone number
v_mobile your mobile phone number
w_vmail your voice mail
to MAILBOX MENU
"current number"
No caller ID
to TELEPHONE NUMBERS MENU
Caller ID
Prompt: v_your_avail Your incoming calls will be transferred to $caller.id.number.
confirm.phr
yes
no
Prompt: vr_is_this_okay Is that okay?
Timeout: repeats prompt
NUMBER CONFIRM RECOG
"no"
to TELEPHONE NUMBERS MENU
"yes"
to MAILBO X MENU
Prompt: great Great.

FIG. 4K

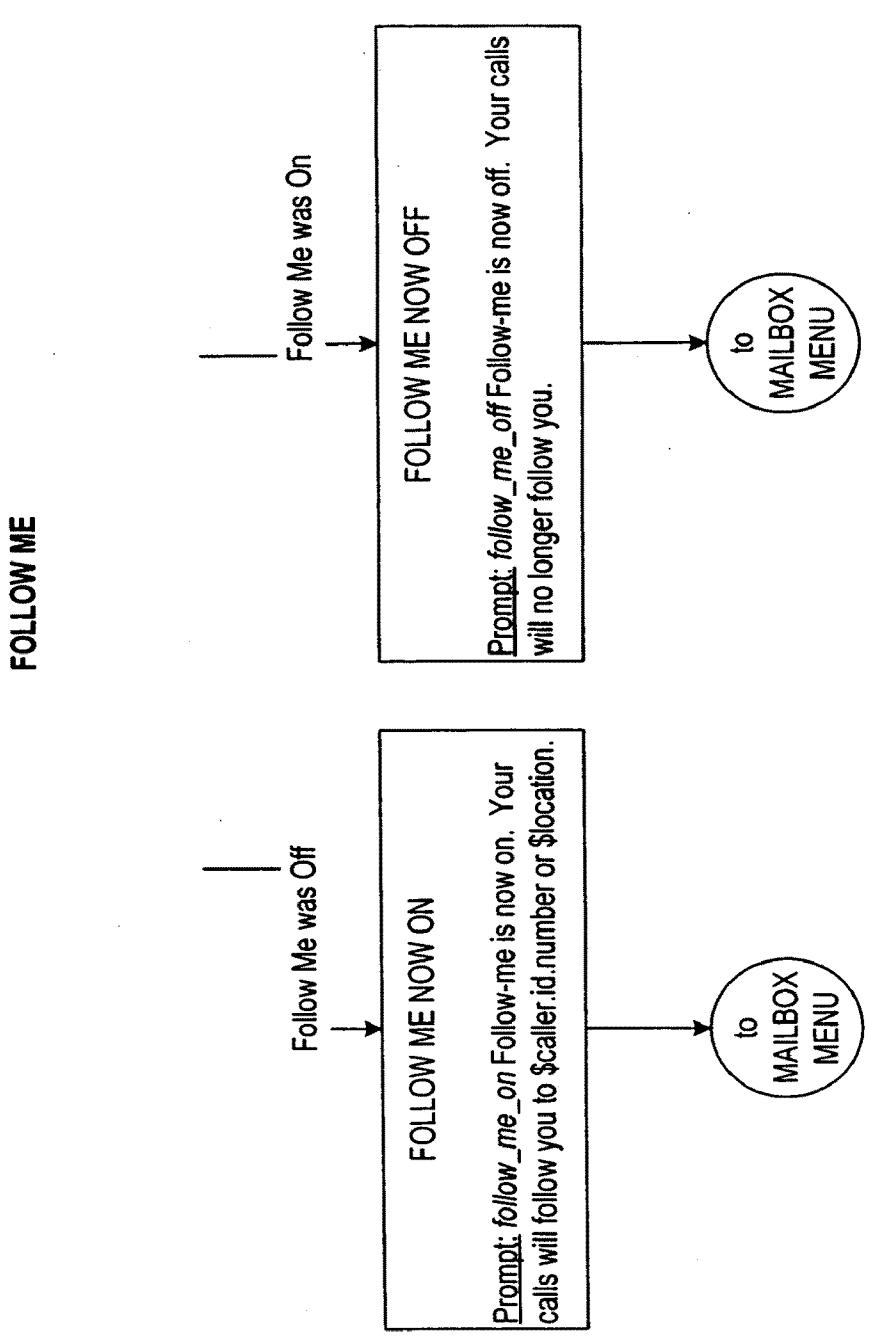

FOLLOW ME
Follow Me was Off
FOLLOW ME NOW ON
Prompt: follow_me_on Follow-me is now on. Your calls will follow you to $caller.id.number or $location.
to MAILBOX MENU
Follow Me was On
FOLLOW ME NOW OFF
Prompt: follow_me_off Follow-me is now off. Your calls will no longer follow you.
to MAILBOX MENU

FIG. 4L

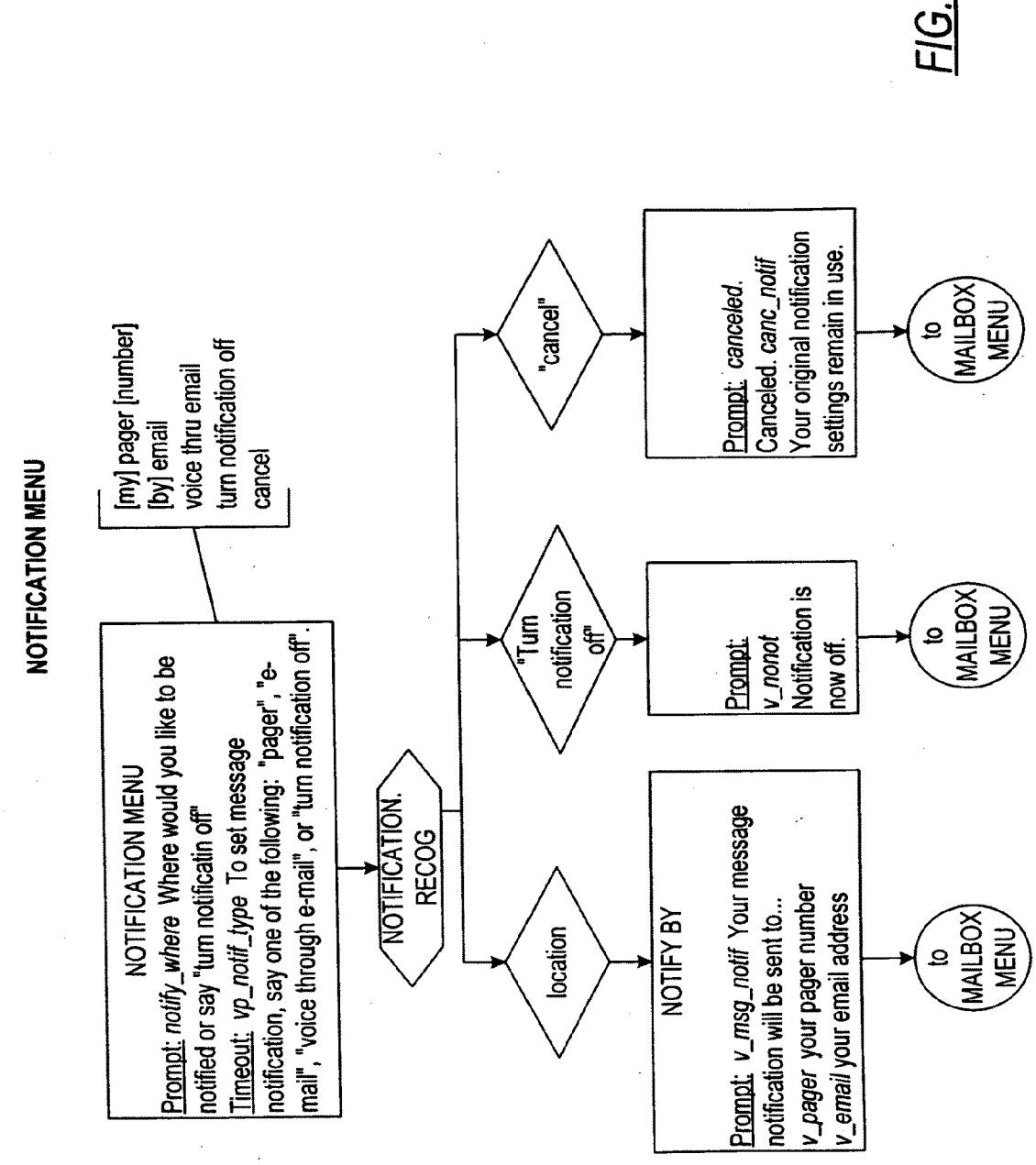

NOTIFICATION MENU
NOTIFICATION MENU
Prompt: notify_where Where would you like to be notified or say "turn notificatin off"
Timeout: vp_notif_type To set message notification, say one of the following: "pager", "e-mail", "voice through e-mail", or "turn notification off".
[my] pager [number]
[by] email
voice thru email
turn notification off
cancel
NOTIFICATION. RECOG
location
NOTIFY BY
Prompt: v_msg_notif Your message notification will be sent to... v_pager your pager number v_email your email address
to MAILBOX MENU
"Turn notification off"
Prompt: v_nonot Notification is now off.
to MAILBOX MENU
"cancel"
Prompt: canceled. Canceled. canc_notif Your original notification settings remain in use.
to MAILBOX MENU

FIG. 4M

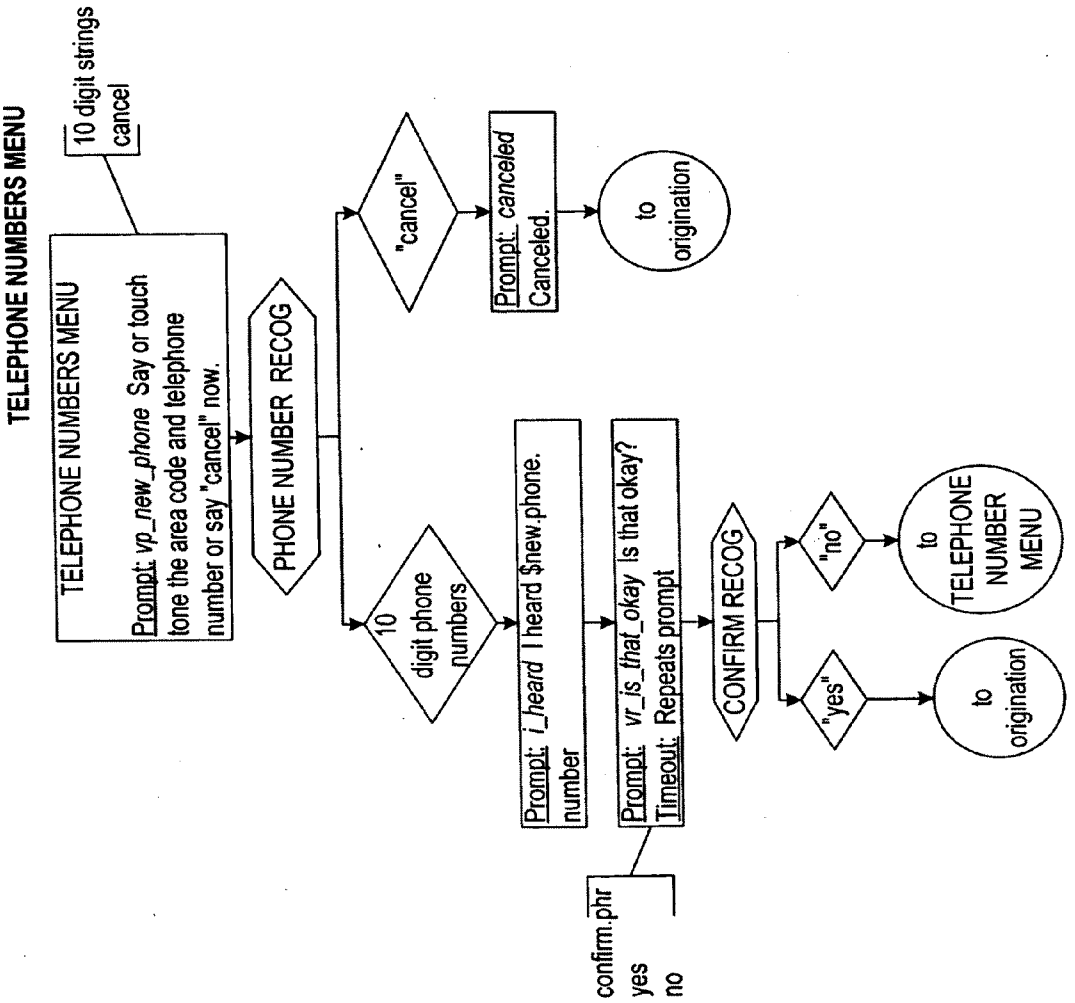

TELEPHONE NUMBERS MENU
TELEPHONE NUMBERS MENU
Prompt: vp_new_phone Say or touch tone the area code and telephone number or say "cancel" now.
10 digit strings
cancel
PHONE NUMBER RECOG
10 digit phone numbers
"cancel"
Prompt: canceled Canceled.
to origination
Prompt: i_heard I heard $new.phone. number
Prompt: vr_is_that_okay Is that okay? Timeout: Repeats prompt
confirm.phr
yes
no
CONFIRM RECOG
"no"
to TELEPHONE NUMBER MENU
"yes"
to origination

FIG. 4N

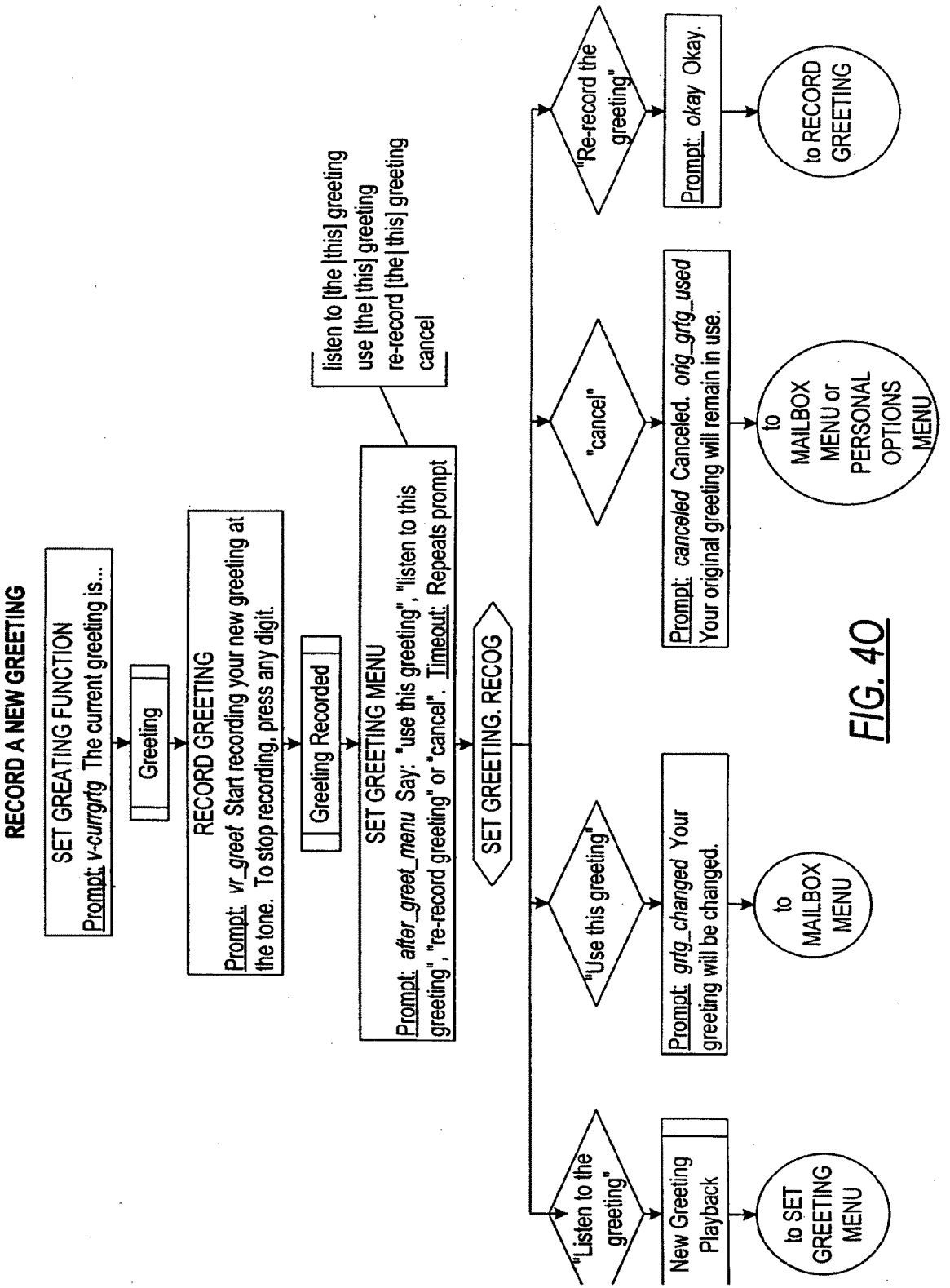

RECORD A NEW GREETING
SET GREATING FUNCTION
Prompt: v-currgrtg The current greeting is...
Greeting
RECORD GREETING
Prompt: vr_greet Start recording your new greeting at the tone. To stop recording, press any digit.
Greeting Recorded
SET GREETING MENU
Prompt: after_greet_menu Say: "use this greeting", "listen to this greeting", "re-record greeting" or "cancel". Timeout: Repeats prompt
listen to [the |this] greeting
use [the | this] greeting
re-record [the | this] greeting
cancel
SET GREETING. RECOG
"Listen to the greeting"
New Greeting Playback
to SET GREETING MENU
"Use this greeting"
Prompt: grtg_changed Your greeting will be changed.
to MAILBOX MENU
"cancel"
Prompt: canceled Canceled. orig_grtg_used Your original greeting will remain in use.
to MAILBOX MENU or PERSONAL OPTIONS MENU
"Re-record the greeting"
Prompt: okay Okay.
to RECORD GREETING

FIG. 4O

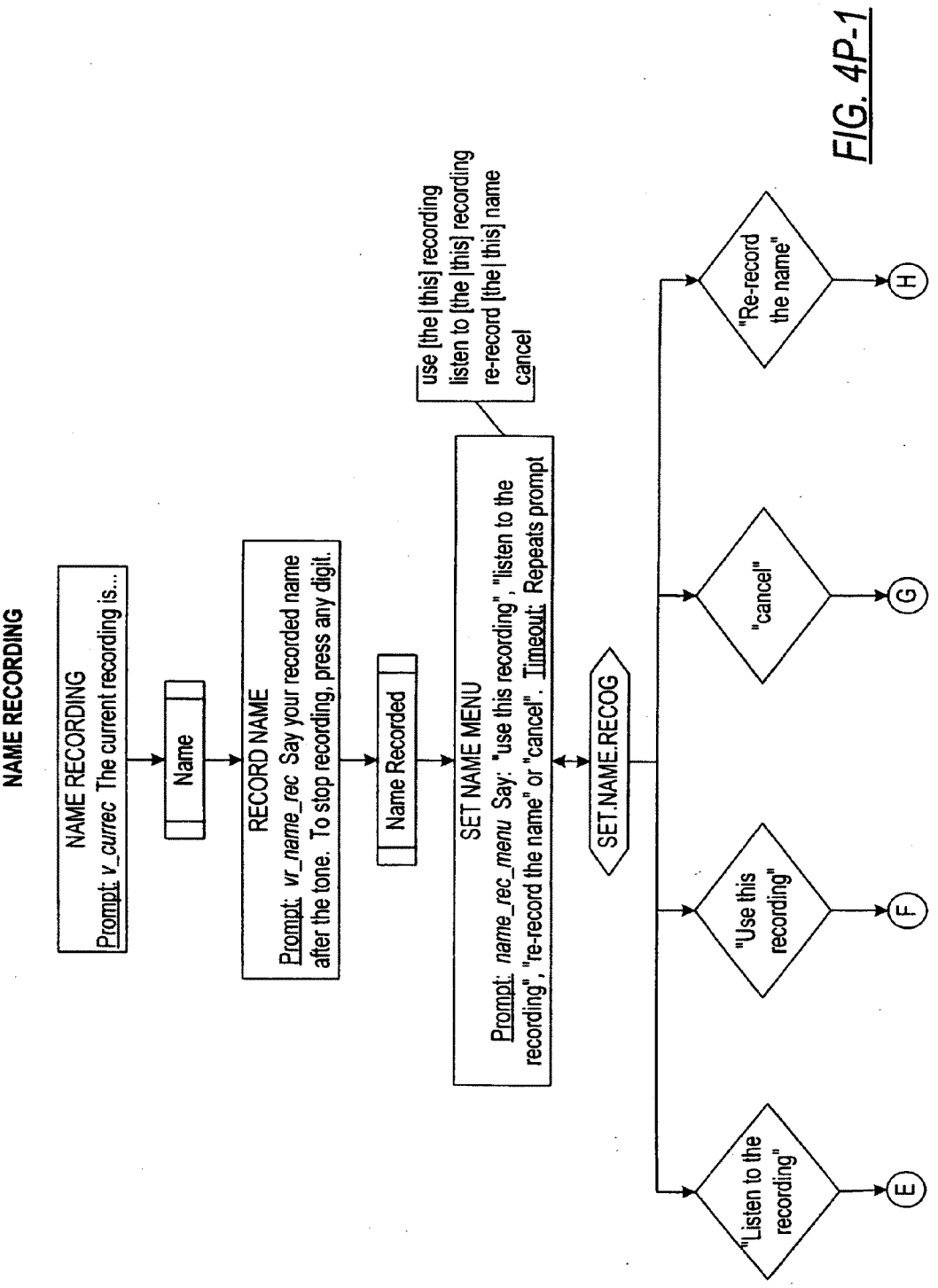

NAME RECORDING
NAME RECORDING
Prompt: v_currec The current recording is...
Name
RECORD NAME
Prompt: vr_name_rec Say your recorded name after the tone. To stop recording, press any digit.
Name Recorded
SET NAME MENU
Prompt: name_rec_menu Say: "use this recording", "listen to the recording", "re-record the name" or "cancel". Timeout: Repeats prompt
use [the | this] recording
listen to [the | this] recording
re-record [the | this] name
cancel
SET.NAME.RECOG
"Listen to the recording"
"Use this recording"
"cancel"
"Re-record the name"
E
F
G
H

FIG. 4P-1

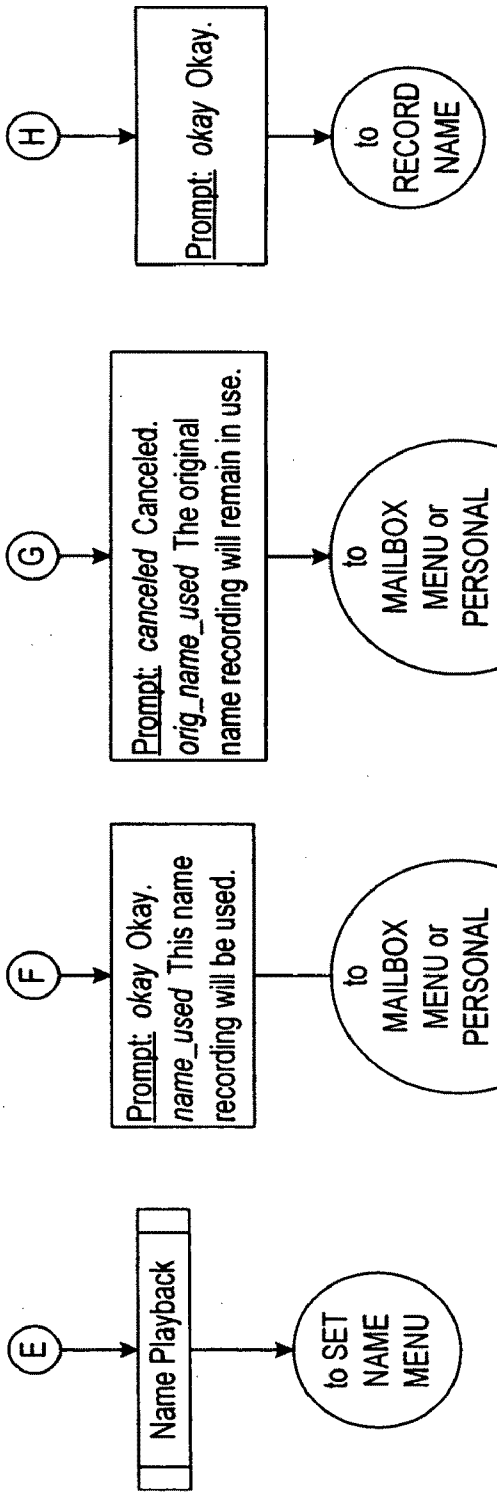

E
Name Playback
to SET NAME MENU
F
Prompt: okay Okay. name_used This name recording will be used.
to MAILBOX MENU or PERSONAL OPTIONS MENU
G
Prompt: canceled Canceled. orig_name_used The original name recording will remain in use.
to MAILBOX MENU or PERSONAL OPTIONS MENU
H
Prompt: okay Okay.
to RECORD NAME

FIG. 4P-2

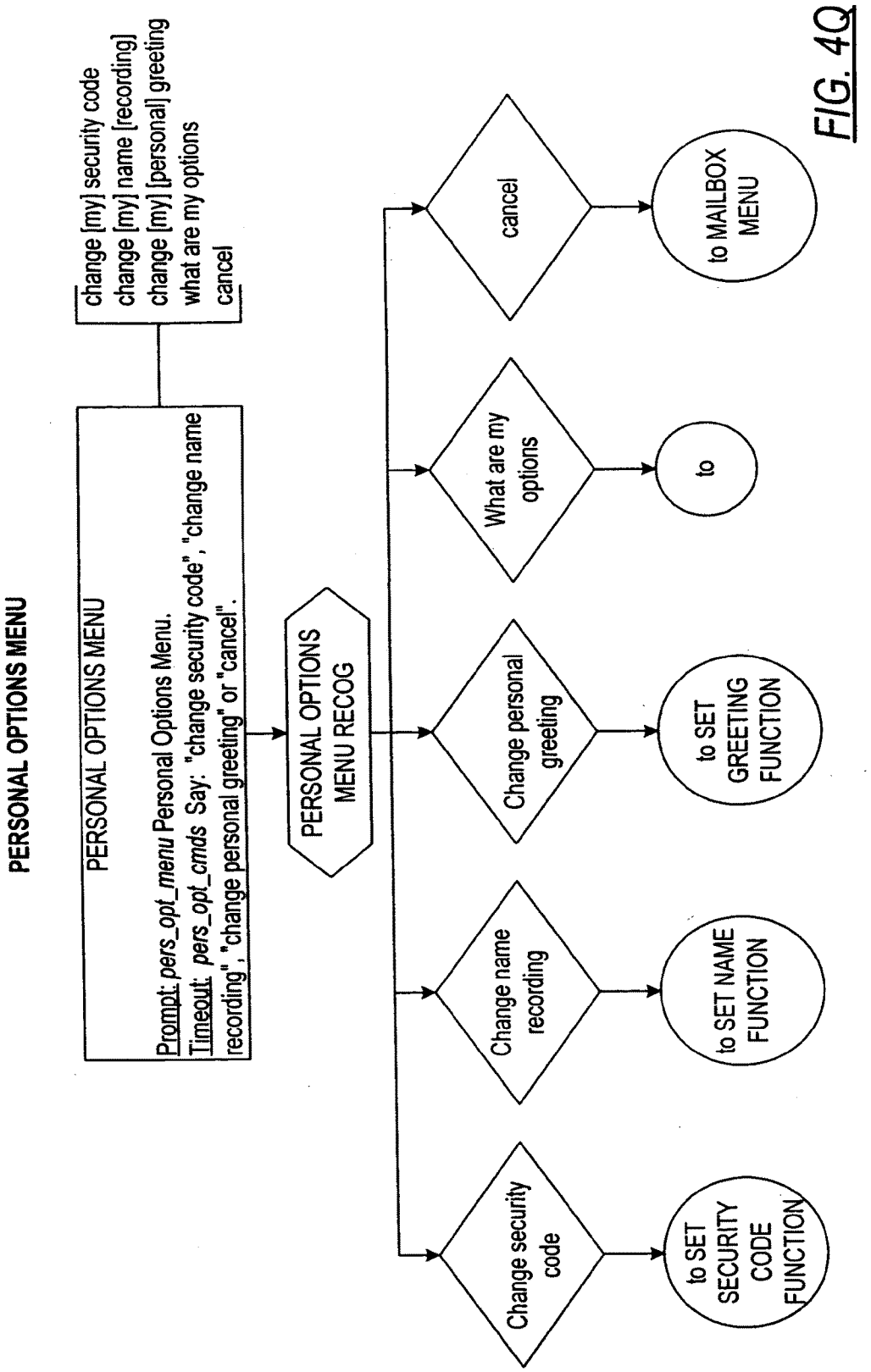

PERSONAL OPTIONS MENU
PERSONAL OPTIONS MENU
Prompt: pers_opt_menu Personal Options Menu.
Timeout: pers_opt_cmds Say: "change security code", "change name recording", "change personal greeting" or "cancel".
change [my] security code
change [my] name [recording]
change [my] [personal] greeting
what are my options
cancel
PERSONAL OPTIONS MENU RECOG
Change security code
to SET SECURITY CODE FUNCTION
Change name recording
to SET NAME FUNCTION
Change personal greeting
to SET GREETING FUNCTION
What are my options
to
cancel
to MAILBOX MENU

FIG. 4Q

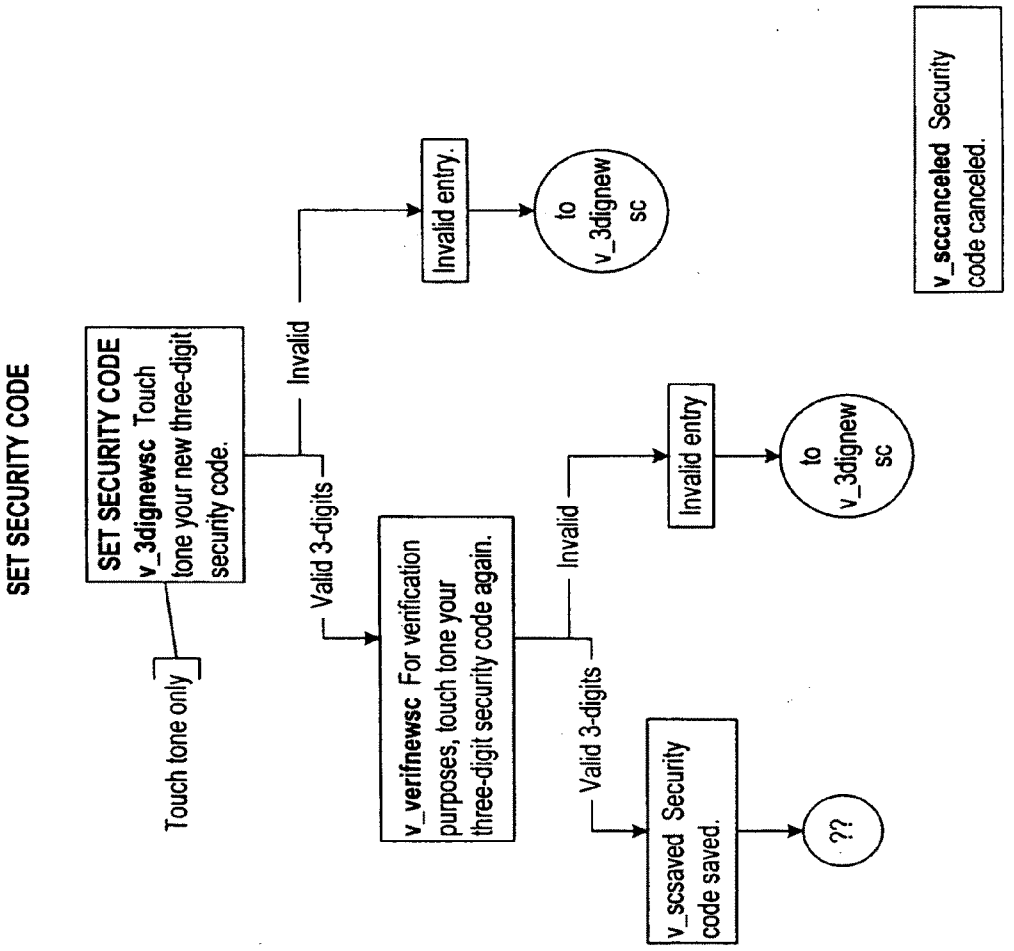

SET SECURITY CODE
SET SECURITY CODE v_3dignewsc Touch tone your new three-digit security code.
Touch tone only
Invalid
Invalid entry.
to v_3dignew sc
Valid 3-digits
v_verifnewsc For verification purposes, touch tone your three-digit security code again.
Invalid
Invalid entry
to v_3dignew sc
Valid 3-digits
v_scsaved Security code saved.
??
v_sccanceled Security code canceled.

FIG. 4R

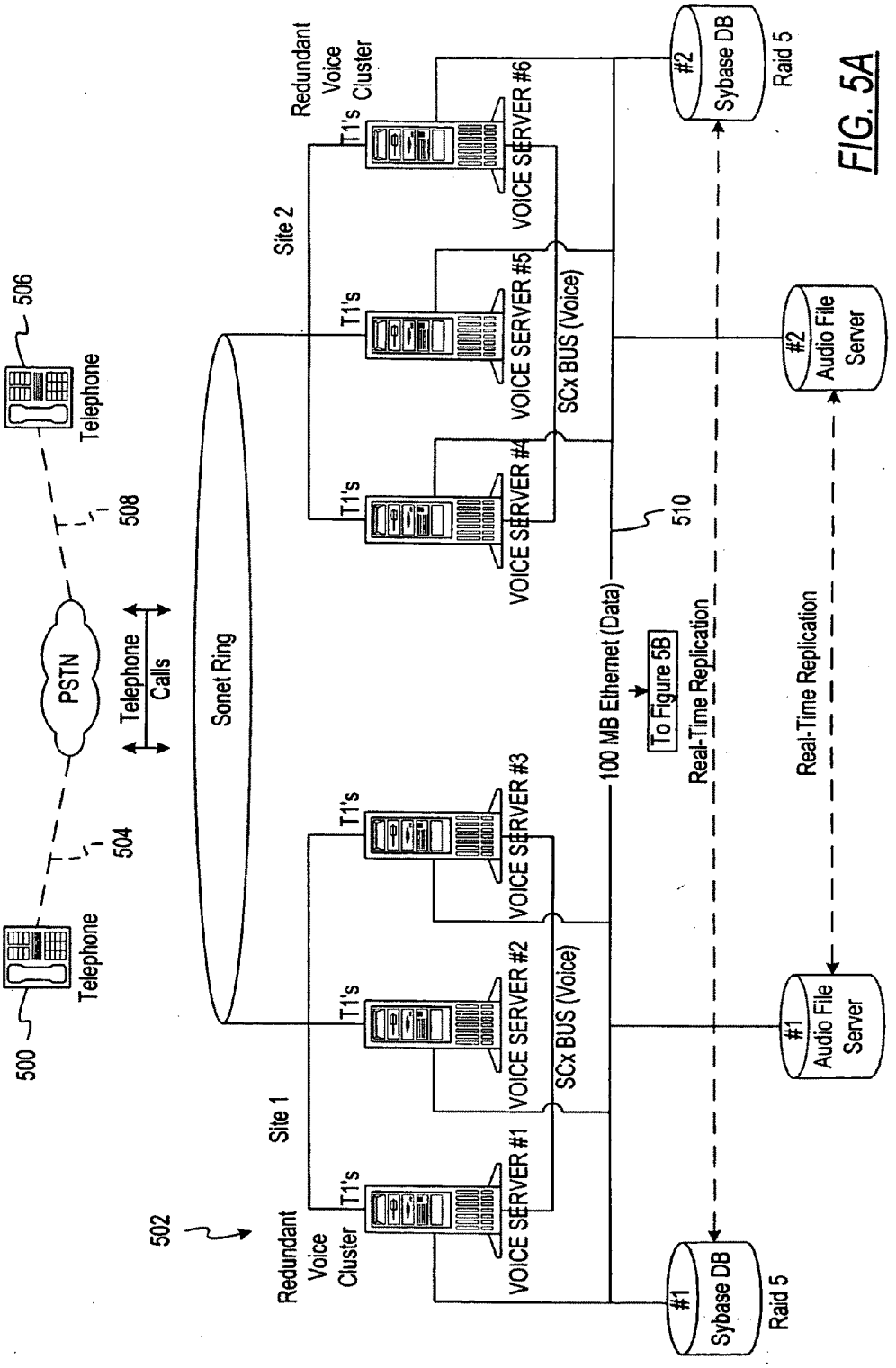

Telephone
506
Telephone
508
PSTN
Telephone Calls
504
Telephone
500
Sonet Ring
502
Site 1
Redundant Voice Cluster
T1's
VOICE SERVER #1
VOICE SERVER #2
VOICE SERVER #3
SCx BUS (Voice)
Site 2
Redundant Voice Cluster
T1's
VOICE SERVER #4
VOICE SERVER #5
VOICE SERVER #6
SCx BUS (Voice)
100 MB Ethernet (Data)
510
To Figure 5B
Real-Time Replication
Real-Time Replication
#1 Sybase DB Raid 5
#2 Sybase DB Raid 5
#1 Audio File Server
#2 Audio File Server

FIG. 5A

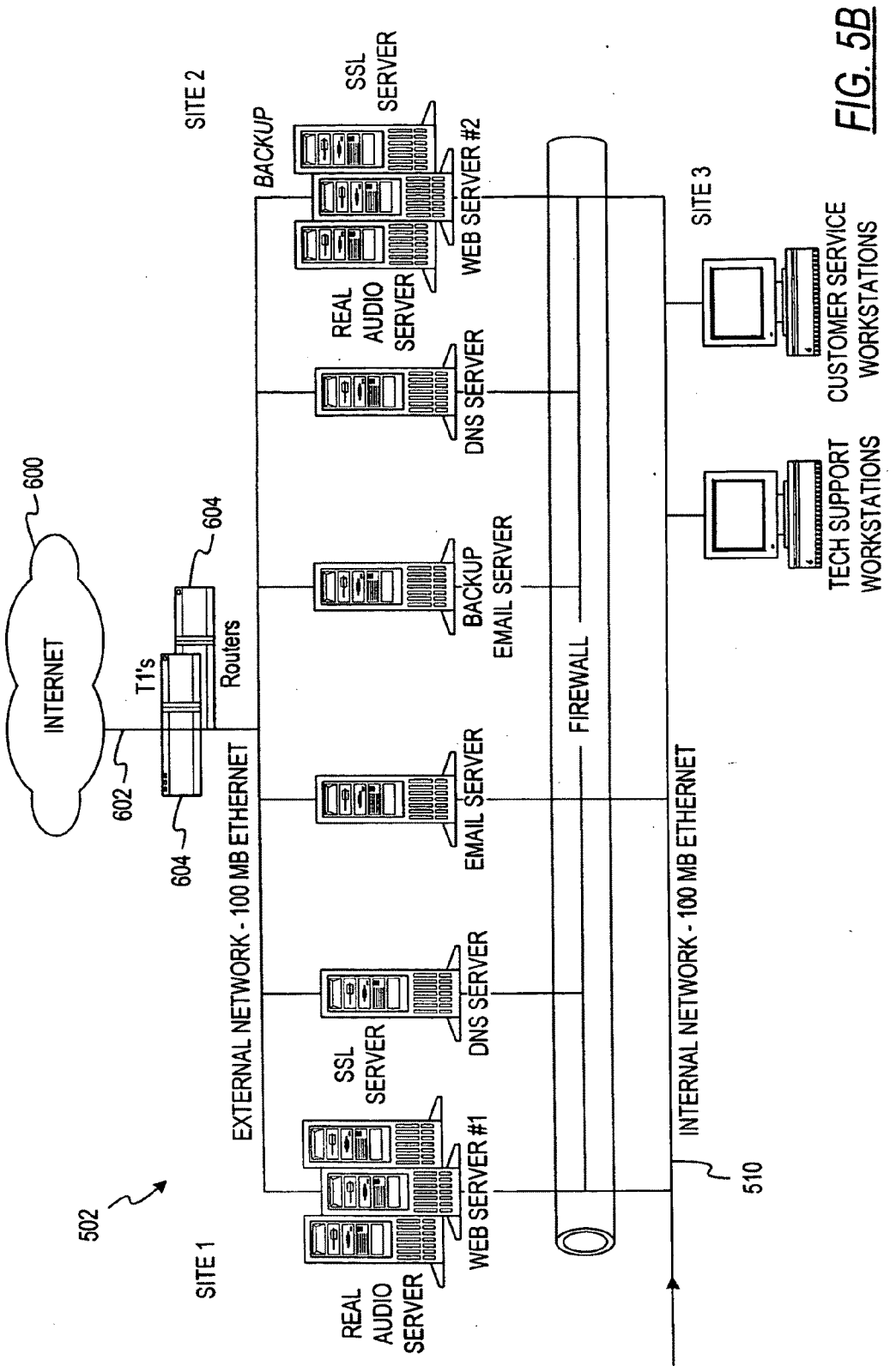

INTERNET
600
602
604
T1's
Routers
604
EXTERNAL NETWORK - 100 MB ETHERNET
SITE 1
502
REAL AUDIO SERVER
WEB SERVER #1
SSL SERVER
DNS SERVER
EMAIL SERVER
BACKUP EMAIL SERVER
REAL AUDIO SERVER
WEB SERVER #2
SSL SERVER
BACKUP
SITE 2
FIREWALL
INTERNAL NETWORK - 100 MB ETHERNET
510
SITE 3
TECH SUPPORT WORKSTATIONS
CUSTOMER SERVICE WORKSTATIONS

FIG. 5B

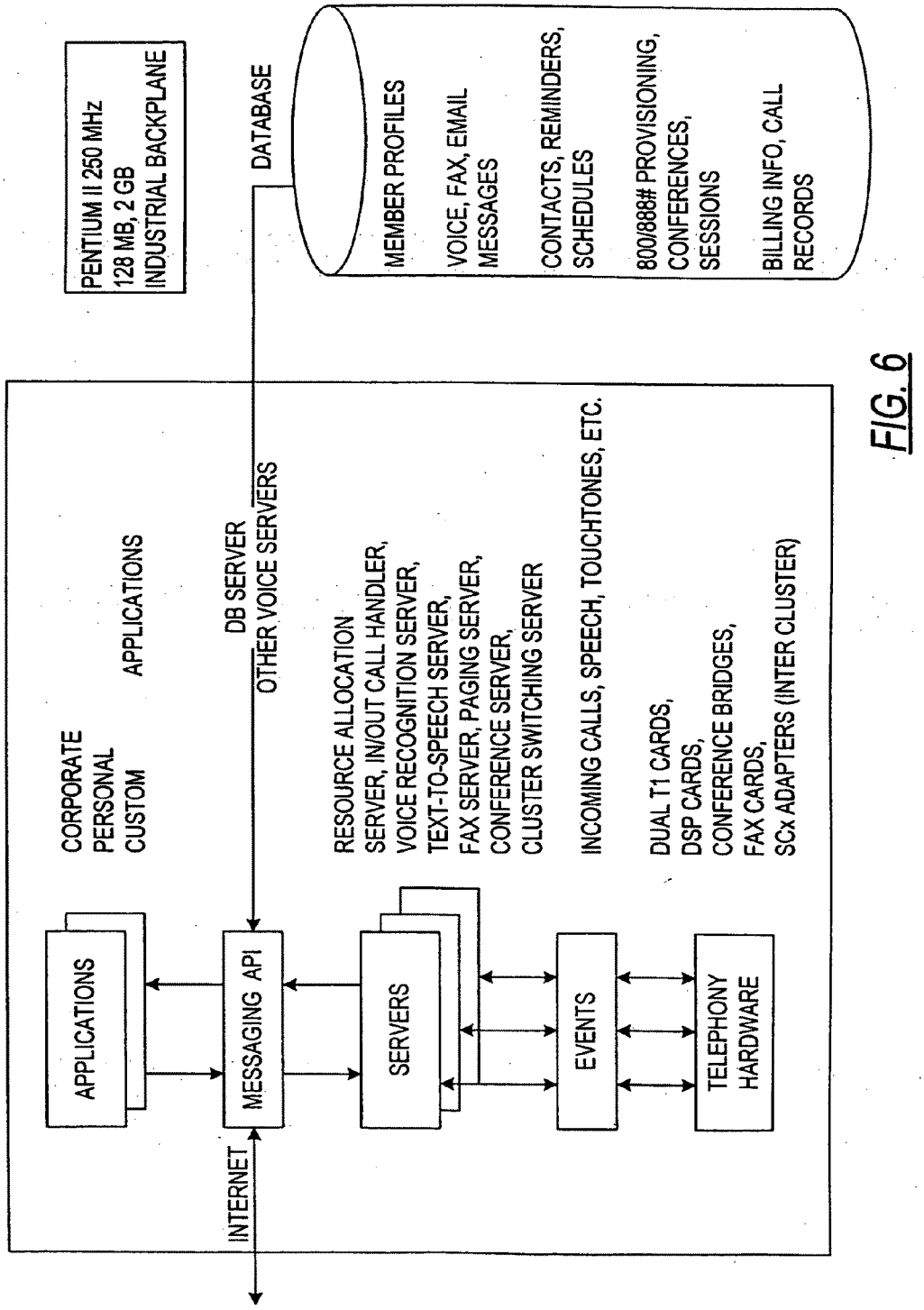

PENTIUM II 250 MHz
128 MB, 2 GB
INDUSTRIAL BACKPLANE
DATABASE
MEMBER PROFILES
VOICE, FAX, EMAIL MESSAGES
CONTACTS, REMINDERS, SCHEDULES
800/888# PROVISIONING, CONFERENCES, SESSIONS
BILLING INFO, CALL RECORDS
CORPORATE
PERSONAL
CUSTOM
APPLICATIONS
DB SERVER
OTHER VOICE SERVERS
RESOURCE ALLOCATION
SERVER, IN/OUT CALL HANDLER,
VOICE RECOGNITION SERVER,
TEXT-TO-SPEECH SERVER,
FAX SERVER, PAGING SERVER,
CONFERENCE SERVER,
CLUSTER SWITCHING SERVER
INCOMING CALLS, SPEECH, TOUCHTONES, ETC.
DUAL T1 CARDS,
DSP CARDS,
CONFERENCE BRIDGES,
FAX CARDS,
SCx ADAPTERS (INTER CLUSTER)
APPLICATIONS
MESSAGING API
SERVERS
EVENTS
TELEPHONY HARDWARE
INTERNET

FIG. 6

COMPUTER, INTERNET AND TELECOMMUNICATIONS BASED NETWORK

[0001] This application is a continuation of U.S. application Ser. No. 09/033,355 and titled "Computer, Internet and Telecommunications Based Network," which was filed on Mar. 2, 1998 and is hereby incorporated by reference in its entirety. Application Ser. No. 09/033,335 is a continuation of U.S. Provisional Patent Application Ser. No. 60/040,056, filed Mar. 3, 1997, and which is hereby incorporated by reference in its entirety.

COMPUTER PROGRAM LISTING APPENDIX

[0002] Compact discs containing a computer program listing are attached to this specification. The computer program listing contained on these compact discs is hereby incorporated by reference into the specification. The appendix comprises one compact disk containing the file "program listing.txt", which is 1.09 MB in size. This file was created on Nov. 6, 2002 and contains the source code which was part of the provisional application with Ser. No. 60/040,056, which this application claims priority. One duplicate copy of this compact disc is also included.

FIELD OF THE INVENTION

[0003] The present invention relates to a telecommunication system, and more specifically, the present invention relates to a network system based on internet, computer and telecommunication standards, utilizing internet and computer technology, a graphical user interface, integrated telecommunication applications and interactive voice recognition technology, facilitating the computation and telecommunication of voice and electronic data.

BACKGROUND OF THE INVENTION

[0004] Individuals and business people seek to communicate with each other, obtain useful information, interact commercially and entertain themselves in an increasingly mobile society. In order to fulfill these needs, one requires the ability to send and receive messages, access information and entertainment, conduct business transactions, organize daily schedules and stay in touch with homes and offices from almost anywhere, at any time, as easily as making a telephone call.

[0005] Continued demand for products and services that address these needs is evidenced by the increasing number of electronic devices, and the explosive growth of the internet and network services. Advances in wireless telecommunication technologies led to the development of such devices as personal digital assistants, and enabled the growth of paging and cellular telephone networks. Devices such as notebook and sub-notebook computers with modems (both wireline and wireless) have allowed mobile professionals to connect to their PCs from almost any location, as well as to access on-line information and electronic mail services while traveling worldwide.

[0006] In addition, communication and information needs have stimulated the growth of the internet, on-line networks and corporate intranets. These networks now host a variety of services such as e-mail, database searching, conferencing, electronic commerce, games, software libraries and electronic newspapers and magazines. However, despite the proliferation of communication devices and the development of the internet, on-line networks and corporate intranets, significant barriers remain to fulfilling user needs for access to and management of personal, professional and public information.

[0007] The hardware designs and software technologies which enable today's communication are complex. Information, though widely available, can be either inaccessible or accessible only by navigating through a host of phone systems, operating system platforms, databases and networks. As a result significant amounts of time and effort are required of those who use and depend on these devices, networks and services to communicate and obtain information.

[0008] The problem of accessing and processing all of the information available from communication devices, networks and services is particularly acute for mobile business professionals. Today's mobile professional, working out of the home or small office, may have a cellular phone, a pager, a computer, a fax machine, an electronic mailbox on the internet, and a voice-mail service. Whether on the road, in a plane or at the office, success for the mobile professional depends in large part on the ability to easily and quickly access, sort through and respond to the messages delivered to each of these communication devices, and to obtain information necessary to the conduct of business from proliferating networks and services.

SUMMARY OF THE INVENTION

[0009] The present invention is a network system, which is based on internet, computing and telecommunications standards, utilizing computer and internet technology, an innovative graphical user interface, integrated communication applications and interactive voice recognition technology. The present invention is a unified messaging service which will be accessible from any standard communication device (telephone, computer or internet), and will give the user intuitive voice command of personal, professional and public information.

[0010] This unified messaging service is a useful tool to those whose time and resources are limited and for whom communication is critical, such as mobile business professionals in the small office, home office market. The mobile business professional must maintain access to personal and professional information and developments, respond to customers and communicate with colleagues, family and friends at any time and from any location. The unified messaging service is designed to meet these objectives by offering a single point of access to all communications, integrated with personal information management tools and customized public content delivery.

[0011] Small office, home office professionals, most of whom do not have access to dedicated information management systems or the benefit of administrative support staff, may derive unique value from this shared network solution.

BRIEF DESCRIPTION OF THE DRAWINGS

[0012] Preferred embodiments of the invention are explained below with references to the accompanying drawings in which:

[0013] FIG. **1** is a functional block diagram of the hardware platform embodying the present invention;

[0014] FIG. **2** is a functional block diagram of the internet platform embodying the present invention;

[0015] FIG. **3** is a general application flow chart embodying the present invention; and

[0016] FIG. **4**A-**4**R are detailed application flow charts embodying the present invention.

[0017] FIG. **5**A-**5**B are the functional block diagrams for the computer telephony platform and network architecture embodying the present invention.

[0018] FIG. **6** is the software architecture embodying the present invention.

DETAILED DESCRIPTION OF THE PREFERRED EMBODIMENTS

[0019] The present invention is a network system that works, among other things, as your voice-mail system, using an 800/888 number that receives all your calls, faxes, pages and e-mail. The system provides a contact database facilitating the placing of calls, screening of calls, and tracking you down wherever you are.

[0020] Regardless of how the message was transmitted, every message is delivered to you through a single source, whether telephone, internet or computer. You can also review your latest call records and billing information and change or add to your phone listings, contact numbers and service preferences through this source.

[0021] The present invention utilizes a fault resilient redundant system, residing in two separate sites. There are multiple T-3 (45 Mbps) facilities going into each of those sites. Both sites sit on a fiber-channel fiber optic loop, which is theoretically "self healing" in case of a disaster and can route a phone call to either facility, depending on where the route is broken.

[0022] The configuration at each site is a cluster of servers. FIG. **1** is a block diagram of the hardware platform showing one of these clusters **10** which consists of some voice servers **12**, some mirrored Sybase database servers **14**, and some web servers **16** where the web access and services are located. The computers (not shown) are 200 MHz Intel-based **19**" rack-mount servers running a combination of Solaris and SCO UNIX operating systems. The voice server **12** includes certain functions, such as telephony, automatic speech recognition, text-to-speech, conferencing, etc. Subscribers are connected to these clusters by either normal telephone connections or by internet connections.

[0023] Each cluster can serve about 10,000 customers. The system is open-ended allowing for the addition of subscribers as needed. Further the design facilitates the easy addition of hardware and software.

[0024] FIG. **2** shows a block diagram of the internet platform. The system may be accessed via the internet instead of a normal telephone connection (voice or touchtone). The subscriber can access the web server **18** through an internet firewall **20**. The subscriber merely enters the system's web site and then can access his account through a security program. Once in his account, the subscriber can access various features such as playing voice mail, reading e-mail and faxes, managing contacts and schedules, among other services. Further, the system can provide a private line **22** for corporate and other subscribers which can enter the web server **18** through an extranet firewall **24**.

[0025] FIG. **3** shows a general application flow chart for an incoming call **26** to the system. The incoming call **26** whether by telephone connection or internet connection is directed either to customer service **28** (through an automatic call distribution-ACD), fax **30**, call routing **32**, or conference **34** function. Call routing **32** is determined by the incoming call control **38**, previously set by the subscriber, and can include a message and notification to the subscriber, call screening, call waiting, and the transfer of the call. Further, if the incoming call **26** is a subscriber **36**, the system will provide additional functions to the subscriber, such as controlling incoming calls **38**, including call screening, call waiting and transferring calls. The subscriber may set priorities to certain callers allowing only certain callers to reach the subscriber while all others are sent to voice-mail to record a message for playback later.

[0026] Also, the subscriber will be able to manage **40** his account. Managing **40** includes setting the options for telephone calls **42**, setting the options for sending and receiving messages, faxes and e-mail **44**, managing the database containing the subscriber's contacts and other lists **46**, and setting the parameters for outgoing call and setting up conferences **48**.

[0027] FIGS. **4**A through **4**R show a more detailed flowchart of the present invention. FIG. **4**A shows the First Menu encountered by a subscriber or an individual calling the system's **800** telephone number. The caller is explained the different options and then the system, if requested, attempts to recognize the subscriber or party's name or extension that the caller is trying to reach.

[0028] FIG. **4**B shows the Party Menu which allows the caller to begin the transfer to the party hr is trying to reach, leave a message, or, if the caller is a subscriber, open his mailbox. FIG. **4**C shows the Mailbox Menu which allows the subscriber to listen to his messages, leave a message, make a call (using the database), dial a telephone number, request the system to follow the subscriber, transfer calls, and set other personal options, including the greeting, security codes, etc.

[0029] FIG. **4**D shows the Listen to Messages Menu which allows the subscriber to listen to, forward, or return the message, along with other options such as adding a contact, etc. FIG. **4**E shows the Message Menu such as forwarding the message, providing caller identification information, and updating the contact database.

[0030] FIG. **4**F shows the Return a Call Menu which will attempt to use caller identification to return the call.

[0031] FIG. **4**G shows the Leave a Message Menu which allows the caller to leave a message to the intended party. FIG. **4**H shows the After Record Message Menu which allows the caller to listen to the message, re-record the message, send the message, or cancel the message. FIG. **4**I shows the Make Call Menu which allows the subscriber to dial a number, cancel, or attempt to obtain a contact and transfer to that contact.

[0032] FIG. **4**J shows the Dial a Number Menu which allows the subscriber to dial a telephone number. FIG. **4**K is the Call Transfers Menu which allows the subscriber to transfer incoming calls to either his caller identification, a telephone number, or some other number of choice. FIG. **4**L is the Follow Me Menu which allows the subscriber to receive calls wherever he may be located. FIG. **4**M shows the Notification Menu which allows the subscriber to be notified either by

pager or e-mail in certain situations. FIG. **4**N shows the Telephone Numbers Menu which allows the subscriber to dial a telephone number either by depressing the telephone keypad or speaking the number.

[**0033**] FIG. **4**O shows the Record a New Greeting Menu which allows the subscriber to record, listen, or re-record a greeting which a caller will hear upon reaching the subscriber. FIG. **4**P shows the Name Recording Menu which allows the subscriber to set a particular name in the system. FIG. **4**Q shows the Personal Options Menu which allows the subscriber to change the security code, the name recording, and the personal greeting. FIG. **4**R shows the Set Security Code Menu which allows the subscriber to set a new security code.

[**0034**] FIG. **5**A shows the computer telephony platform in accordance with FIG. **1** in more detail. FIG. **5**B shows the network architecture in accordance with FIG. **2** in more detail. FIG. **6** shows the software architecture utilized in the present invention.

[**0035**] The system provides three ways for the subscriber to handle his communications. First and foremost is the voice recognition software using natural voice recognition (phonemes based), not pattern based as many of the current systems utilize. Therefore, the system does not have to be trained to identify your voice. Second, the subscriber may use the standard telephone touchtones. And third, the subscriber can utilize the internet to access a secure web site.

[**0036**] The system can act as a "follow me" service, capable of tracking you down whether you're at home, the office, airport, in a hotel room, in another office or in your car. The subscriber can configure the system to follow him, or the subscriber can disengage the follow me with a simple command. It also lets you know who is calling before you accept the call.

[**0037**] As for Web connections, you can boot up your computer, sign onto the internet, go through the system's secure web site and play your voice mails, read your e-mail and faxes, manage your contacts and schedule conferences. The website thus becomes a unified-messaging system. If you get a message from somebody who's not in your address book and you want to call him or her, you just go to the web site again and add a contact.

[**0038**] If you're on the secure website and you walk away, the system even times out after a few minutes so no one can sit at your computer and obtain your messages and contacts without your permission.

[**0039**] While on the system's web site, you'll be communicating over the internet via the Secure Socket Layer (SSL), a transport level technology (developed by Netscape) for authentication and data encryption between a Web server and a Web browser. SSL sends data over a "socket," a secure channel at the connection layer existing in most TCP/IP applications.

[**0040**] The system also utilizes text-to-speech so you can listen to your e-mail over the phone, or the subscriber can display his e-mail on his computer where he can reply to it and send e-mail to his contacts with audio (using audio attachments). Further, using Vosaic's internet audio program, the system can deliver voice mail via streaming audio over the internet using Java without a plug-in.

[**0041**] During a session on the web site, using text-to-speech, you could type in a name like "Harry Newton" and his phone number so that, when you eventually use the system the phone and you say, "call Harry Newton at the office," the system will recognize the name and dial the number you keyed in previously. Based on what you input now ("Harry Newton") is what the system utters later and compares the phonemes of what you input now to what you'll be speaking later.

[**0042**] The system also facilitates conference calls by allowing the subscriber two ways in which to set up a conference call. First, inbound conference calls can be set up for 32 people if you've registered a pre-scheduled meeting time. Subscribers call in to the system to get an assigned conference number. The conference participants then call in and asks to "join the conference". The system then asks for a conference number, which is entered via touchtones. All the parties are then connected and the conference begins.

[**0043**] Second, outbound conference calls can also be set up, even if you're "on the fly." Whether you're at your desk, at a hotel or even a phone booth, you can simply dial up and tell the system to "set up a conference call." Then you only have to dial up all the various parties to connect each of them to your conference.

[**0044**] The present invention is a compilation of hardware and software, including voice processing using Dialogic's SCSA extended bus and board-level resources, Purespeech speech recognition running on Dialogic Antares boards, and database management using Sybase System **10** and **11**. Further, the system utilizes data networking, particularly TCP/IP and distributed systems, object-oriented design and programming, multi-processing with Intel hardware, SCO UNIX and Solaris operation systems, Java and JavaScript languages, US and international long distance protocols, internet and web protocols, credit and payment processing, a help desk, customer service system, and network and service management.

[**0045**] For the voice and fax processing servers, the system relies on Dialogic hardware, including, voice boards, Antares cards (for speech recognition and fax) and digital switching cards. The system also uses the SCSA extended bus. The Purespeech ASR algorithms, which are used for speech recognition, run on the Antares cards plugged into the computers running the Solaris OS.

[**0046**] The system combines state-of-the-art speech recognition, computer and telephony technology. Along with the ability to recognize an extensive set of simple, intuitive, speaker-independent speech commands and respond by performing a wide variety of complex tasks. Each subscriber has their own secure Web page on which all the features can be accessed, allowing almost every aspect of the present invention to be maintained on-line.

[**0047**] Subscribers can use voice commands to (1) build, edit and manage their contact lists, (ii) review, play back, read, reply to and/or reroute voice mail and e-mail, (iii) schedule conference calls with 800 number access, and (iv) maintain an itemized calling log listing all calls with a running total of all charges.

[**0048**] The system also transfers calls as instructed or upon command will follow the subscriber according to predetermined contact numbers for office, home, cellular, pager or other designated locations. Every time a subscriber calls in,

the system logs the originating number for the inbound call and then uses that number to re-contact the subscriber.

[0049] In accordance with the present invention, every subscriber or participant is given their own 800 number that is then used by all contacts calling in to the subscriber. The 800 number can also be used in scheduling conference calls for up to 32 participants on a given day and time. Each participant uses the 800 number and calls in to join the conference call, thereby effecting call conferencing for a fraction of the cost otherwise incurred.

[0050] Basically, the system provides a unified solution to the many varied communications and messaging devices used daily by mobile professionals and active consumers. With simple voice commands, subscribers can easily access and respond to all of their communications and messaging media in the same session.

[0051] The system takes inbound calls and contacts the subscriber, using call transfer or follow me features, and advises the subscriber of the call, the number of the calling party and/or the callers identity from the subscribers contact list. If the subscriber is on the telephone, the system will whisper the pending call information, giving the subscriber the option of taking the call or sending the caller into voice mail. The subscriber can also set priorities for certain calls which the system will follow, permitting selected calls to be put on call waiting, transferred or directed through call forwarding, while other calls are direct to voice mail messaging. Subscribers can conveniently make calls from their contact list by voice commands giving the contacts name and, if applicable, the location to be called (i.e. "home,""office, ""cellular," etc.)

[0052] Subscribers can access and play back their voice mail from any telephone or from their personal Web page. Voice mail messages can be saved, retrieved, deleted or rerouted to other individuals, groups or broadcast and voice responses can be returned immediately to the caller. Also, e-mail messages can be viewed on the subscriber's personal home page or the system will read the e-mail to the subscriber from any telephone using text-to-speech technology. Subscribers can immediately respond to e-mail with voice messages, marking, saving or deleting messages during the same session. E-mail messages can also be sent to any fax machine. Further, subscribers can immediately respond to faxes rather than waiting to retrieve copies and delay responses. Incoming faxes are received, the subscriber is notified of the arrival of the new fax, the fax can then be stored for later viewing and/or redirected to any fax machine or e-mail address from the subscribers contact list or any other number. Subscribers can also be notified of any incoming communications and messages by pager, whether an inbound call, voice mail, e-mail or fax.

[0053] The system will automate conference calling and eliminate the need for conference call operators or complex, confusing PBX systems. The subscriber remains in control and can add or drop callers, mute the call and otherwise control all aspects of the conference call. By pre-scheduling conference calls by date and time, subscribers can notify participants, giving them an 800 number to call and join the conference. Up to 32 participants can be included in the present conference call feature.

[0054] The system further acts as an efficient secretary retaining complete contact logs of all call statistics. The subscriber maintains the contact list using simple voice or keyboard commands and can establish contact groups for broadcast communications and setting up conference calls. The contact database is easily accessible to all features.

[0055] The system includes a feature which enables subscribers to retrieve on demand or at predetermined intervals selected information from the internet or on-line service providers, allowing subscribers to establish "filter and forward" criteria specifying the type of information desired. A search engine will then retrieve the requested information, transmit the information to the system platform and notify the subscriber by page, telephone or other desired means. The system will provide direct access to news, weather, sports, financial, travel and other custom content directly from a computer or any telephone. The subscriber will then access the information by all available options, including text-to-speech capabilities.

[0056] The system provides long distance and international calling over the internet through the subscriber's personal home page, thereby significantly reducing the costs of long distance and international calling. The system further provides video conferencing features.

[0057] It is to be understood that the form of this invention is merely a preferred embodiment. Various changes may be made in the function and arrangement of parts; equivalent means may be substituted for those illustrated and described; and certain features may be used independently from others without departing from the spirit and scope of the invention as defined in the following claims.

1. A system for sending and receiving communications to and from a plurality of users via different modes of communication and for managing communications by a plurality of system users, said system comprising:

a telephone interface configured to allow any of said plurality of system users to send and receive telephone communications;

an Internet or internet interface configured to allow any of said plurality of system users to send and receive Internet or internet communications;

a first plurality of predetermined rules for each of said system users for managing incoming telephone communications received via said telephone interface;

a second plurality of predetermined rules for each of said system users for managing incoming Internet or internet communications received via said internet interface;

a first plurality of computer instructions for implementing said first plurality of predetermined rules;

a second plurality of computer instructions for implementing said second plurality of predetermined rules;

a third plurality of computer instructions for implementing user telephone instructions received from any of said plurality of system users via said telephone interface or said Internet or internet interface for establishing or revising said user's first plurality of predetermined rules or for real-time managing of an incoming telephone communication by said user;

a fourth plurality of Internet or internet instructions received from any of said plurality of system users for establishing or revising said second plurality of predetermined rules;

a speaker-independent speech recognition device configured to receive natural voice user instructions from any of said plurality of system users, said speech recognition device using phoneme comparisons to convert said natural voice user instructions into said third plurality of instructions or into said fourth plurality of instructions; and

a computer for implementing said first plurality of predetermined rules, said second plurality of predetermined rules, said third plurality of computer instructions and said fourth plurality of computer instructions.

2. The system of claim 1 further comprising a third plurality of predetermined rules to manage outbound telephone communications and a fourth plurality of predetermined rules to manage outbound Internet or internet communications.

3. The system of claim 2 further comprising additional stored or real-time telephone instructions to manage outbound telephone communications.

4. The system of claim 2 further comprising additional stored or real-time internet instructions to manage outbound Internet or internet communications.

5. The system of claim 1 wherein said telephone communications are selected from the group consisting of telephone calls, conference calls, facsimile messages, voice mail messages, and text-to-speech messages.

6. The system of claim 1 wherein said Internet or internet communications are selected from the group consisting of telephone calls, conference calls, facsimile messages, text messages, data communications, and voice messages.

7. The system of claim 1 wherein said first plurality of predetermined rules include rules for transferring telephone communications, following me, setting message notifications, and changing personal greetings.

8. The system of claim 1 wherein said natural voice commands are selected from the group consisting of: "transfer my calls,""transfer calls,""transfer my calls to my office,""transfer my calls to my home,""transfer my calls to my mobile,""transfer my calls to my cell phone,""transfer my calls to my cellular phone,""transfer my calls to my voice mail,""transfer my calls to my current number,""transfer my calls,""transfer calls,""transfer calls to my office,""transfer calls to my home,""transfer calls to my mobile,""transfer calls to my cell phone,""transfer calls to my cellular phone,""transfer calls to my voice mail,""transfer calls to my current number,""follow me,""set message notification,""notify me by email,""notify me by pager number,""notify me by pager,""notify me by voice thru email,""notify by email,""notify by pager number,""notify by pager,""notify by voice thru email,""turn notification off,""cancel,""re-record the greeting," re-record this greeting, "re-record greeting,""change my personal greeting,""change my greeting,""change personal greeting," and "change greeting."

9. The system of claim 1 wherein each one of said plurality of system users can be contacted by calling a single system telephone number.

10. The system of claim 9 further comprising an audible prompt generator such that a caller contacting one of said plurality of system users receives at least one audible prompt requesting at least one natural voice command being descriptive of a desired voice-enabled function.

11. The system of claim 10 wherein said natural voice command is selected from the group consisting of: "what are my options,""transfer to the operator,""transfer to operator,""sales and marketing,""technical support,""dial an extension,""dial extension,""reach an extension,""reach extension,""(first and last name),""find him,""find her,""find them,""leave a message,""re-record the message,""re-record message, "re-record this message,""leave message,""send the message,""send this message,""send message,""reach my party,""reach him,""reach her," and "cancel."

12. The system of claim 1 wherein each one of said plurality of system users is assigned a unique telephone number.

13. The system of claim 12 wherein a caller can directly contact each one of said users by dialing said unique user's telephone number.

14. The system of claim 12 further comprising an audible prompt generator configured to generate an audible prompt such that a caller for one of said plurality of system users via said user unique telephone number receives at least one audible prompt requesting at least one natural voice command being descriptive of a desired voice-enabled function.

15. The system of claim 14 wherein said natural voice command is selected from the group consisting of: "what are my options,""transfer to the operator,""transfer to operator,""(first and last name),""find him,""find her,""find them,""leave a message,""re-record the message,""re-record message, "re-record this message,""leave message,""send the message,""send this message,""send message,""reach my party,""reach him,""reach her," and "cancel."

16. The system of claim 1 wherein each one of said plurality of users can be contacted by calling a single system telephone number or a unique telephone number assigned to each user.

17. The system of claim 1 wherein said speech recognition device comprises a Purespeech speech recognition engine.

18. The system of claim 1 wherein said telephone interface comprises a telephony device operatively connected to said speech recognition device.

19. The system of claim 1 further comprising:

a third plurality of predetermined rules for managing outbound communications;

stored instructions for managing outbound communications in accordance with said third set of predetermined rules; and

real-time outbound instructions for managing outbound communications.

20. The system of claim 31 further comprising a telephony device configured to receive instructions from each of said plurality of users to establish or change said stored or real-time telephone instructions for managing incoming communications or to establish or change said stored or real-time outbound instructions for managing outgoing communications, using DTMF signals.

21. The system of claim 31 further comprising an internet device configured to

receive instructions from each of said plurality of users to establish or change said stored or real-time Internet or internet instructions for managing incoming communications or to establish or change said stored or real-time

outbound instructions for managing outgoing communications, using DTMF signals.

22. A method of sending, receiving, and managing communications transmitted through a computer-based telephony system accessible by a plurality of users, said method comprising the steps of:

providing a telephony device;

providing a speaker-independent speech recognition device operatively connected to said telephony device, said speech recognition device configured to receive natural voice commands from said plurality of system users;

providing a first plurality of predetermined rules for managing incoming telephone communications received via a telephone interface configured to allow any of said plurality of system users to send and receive telephone communications;

providing a second plurality of predetermined rules for managing incoming Internet or internet communications received via an Internet or internet interface configured to allow any of said plurality of system users to send and receive internet communications;

receiving at least one natural voice command to establish or change at least one predetermined rule from said first and second plurality of predetermined rules to manage at least one incoming telephone, Internet or internet communication;

analyzing said at least one natural voice command by comparing phonemes of said at least one natural voice command with phonemes for each predetermined rule in said first and second plurality of predetermined rules; and changing or establishing said at least one predetermined rule to manage at least one incoming telephone, Internet or internet communication.

23. The method of claim 22 further comprising the step of providing a third plurality of predetermined rules for managing outbound communications.

24. The method of claim 23 wherein said natural voice command establishes or changes at least one of the third plurality of predetermined rules for managing outbound communications.

25. A system of claim 1 wherein said Internet interface is configured to receive communications from the Internet.

26. The system of claim 1 wherein said internet interface is configured to receive communications from an internet.

27. A method for receiving, sending and managing information between a computer and telecommunications network and a subscriber communication device of a subscriber comprising the steps of:

accessing at least one voice server via a first telephone communication or via a first internet connection, said voice server containing telephony and speaker-independent speech command recognition and execution functions, such that said subscriber can access said voice server via a first telephone connection or via a first internet connection;

manipulating and managing data and/or at least one contact list via said voice server said data and at least one contact list stored in a database operatively connected to said voice server; and

receiving a first message from a second telephone connection or receiving a second message from a second internet connection and transmitting said first message or said second message to said subscriber communication device or transmitting to said subscriber communication device a notification about said first message or said second message via said first telephone connection or said first internet connection, based on commands from said subscriber to said network.

28. The method of claim 27, wherein said voice server receives said second message from an Internet connection.

29. The method of claim 27 wherein said message comprises telephone calls, e-mail, faxes, conference calls, or voicemail.

30. The method of claim 27 wherein manipulating and managing said data and said contact lists comprises creating and modifying at least one contact list.

31. The method of claim 27 wherein manipulating and managing said data and said contact lists comprises utilizing said contact list to place telephone calls, send e-mail, voicemail and faxes, and create or modify conference calls.

32. A method for receiving, sending and managing information between a computer and telecommunications network and a subscriber communication device of a subscriber comprising the steps of:

accessing at least one voice server via a first telephone connection or via a first internet connection, said voice server containing telephony and speaker-independent speech command recognition and execution functions, such that said subscriber can access said voice server via a first telephone communication or via a first internet connection;

manipulating and managing data and/or at least one contact list via said voice server, said data and at least one contact list stored in a database operatively connected to said voice server;

receiving a first message from said subscriber via said first telephone connection;

receiving a second message from said subscriber via said first internet connection; and

transmitting said first message or said second message via a second telephone connection or a second internet connection, based on commands received by said network from said subscriber.

33. The method of claim 32 wherein said voice server receives said second message from an Internet connection.

34. The method of claim 32 wherein said message comprises telephone calls, e-mail, faxes, conference calls, or voicemail.

35. The method of claim 32 wherein manipulating and managing said data and said contact lists comprises creating and modifying at least one contact lists.

36. The method of claim 32 wherein manipulating and managing said data and said contact lists comprises utilizing said contact list to place telephone calls, send e-mail, voice mail and faxes, and create or modify conference calls.

37. A voice-enable system for managing communications transmitted through a network, said system capable of eliciting a response from a user, said response comprising at least one natural voice command, said system comprising:

a computer;

at least one set of stored commands operatively associated with said computer, each said set including a plurality of stored commands each of said commands in said set corresponding to a single voice-enabled function for managing communications, each said command in said set being an alternative intuitive description of said single voice-enabled function;

a voice server operatively connected to said computer and to said network, said voice server configured to compare said natural voice commands to said plurality of stored commands;

a speaker-independent speech recognition device operatively connected to said voice server, said device configured to elicit at least one natural voice command from said user, said voice server being configured to compare said natural voice command to said plurality of stored commands to select said single voice-enabled function, and said computer being configured to execute said selected voice-enabled function corresponding to said natural voice command.

38. The system of claim 37 wherein said speaker-independent speech recognition device is configured to prompt said user with a prompt.

39. The system of claim 38 wherein said prompt is selected from the group consisting of: "What now?", "What are my options?", "What would you like to do?", and "What would you like me to do?"

40. The system of claim 37 wherein said voice-enabled functions are selected from the group consisting of: opening a network communication, creating a network communication, modifying a network communication, playing a network communication, reading a network communication, sending a network communication, forwarding a network communication, saving a network communication, deleting a network communication, screening a network communication, converting a fax communication into an e-mail, storing a network communication for future retrieval, storing a network communication for transmittal, creating a fax message from a computer-generated communication, and retrieving a communication.

41. The system of claim 40 wherein said network communication is selected from the group consisting of: a telephone call, a conference call, a facsimile message, a voicemail message, an e-mail message, and a message containing data.

42. The system of claim 37 wherein the speech recognition device is operatively connected to an internet.

43. The system of claim 37 wherein the speech recognition device is operatively connected to the Internet.

44. The system of claim 37 wherein the speech recognition device is operatively connected to a telephony device.

45. The system of claim 37 wherein said speech recognition device is operatively connected to a file server.

46. The system of claim 37 wherein said speech recognition device is operatively connected to a database server.

47. The system of claim 37 wherein said speech recognition device is operatively connected to a web server.

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