Using Virtual Environments to Prototype Auditory Navigation Displays

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There is a critical need for navigation and orientation aids for the visually impaired. Developing such displays is difficult and time consuming due to the lack of design tools and guidelines, the inefficiency of trial-and-error design, and experimental participant safety concerns. We discuss using a virtual environment (VE) to help in the design, evaluation, and iterative refinement of an auditory navigation system. We address questions about the (real) interface that the VE version allows us to study. Examples include sound design, system behavior, and user interface design. Improved designs should result from a more systematic and scientific method of assistive technology development. We also point out some of the ongoing caveats that researchers in this field need to consider, especially relating to external validity and over-reliance on VE for design solutions.

Key Words: Auditory display—Blind navigation—Sonification

There are millions of people with vision loss in the United States, approximately 10% of whom have no usable vision. The prevalence rises dramatically with age, and by 2010 these numbers will nearly double (DeL'Arzne; 2002; Goodrich, 1997; Leonard, 2002; National Center for Veteran Analysis and Statistics, 1994). As the population of the United States ages, there will continue to be more persons with visual impairments resulting from, for example, glaucoma, macular degeneration, and diabetic retinopathy. For a person with vision loss, the two fundamental tasks of navigating through a space and knowing what is around can be a great challenge. Spatial orientation is the major mobility problem encountered by all individuals with profound vision loss (LaCrow & Weessie, 1994; Welsh & Blasch, 1997), but is especially difficult for people whose amount of vision loss occurs later in life (Leyr & Gordon, 1986; Welsh & Blasch, 1997). This includes a growing sector of the aging workforce. Wayfinding (the ability to find one's way to a destination) is dependent on the ability to remain oriented in the environment in terms of the current location and heading and the direction of a destination. Even highly experienced blind pedestrians exhibit random movement error large enough to occasionally veer into a wall or into a parallel street when crossing an intersection (Guth & LaDuke, 1995). These problems can be compounded when the person is indoors by the lack of external orienting cues such as the sound of traffic, noise from the flow of other pedestrians, or the chirping of birds in a particular tree. Although there has been a great deal of research in the area of electronic travel aids for obstacle avoidance, there has not been comparable research in the development of orientation devices that keep one apprised of both location and heading (Blasch, Wiener, & Welsh, 1997). Thus, there is a critical need for navigation and orientation aids for the visually impaired.

Such navigation aids also may assist people who find navigation problematic because of other conditions that are unrelated to vision, such as poor balance or other mobility difficulties, or to simply forgetting the route or destination due to cognitive impairments. For these individuals, their visual attention is often dominated by other simple tasks, such as maintaining balance or attempting to identify landmarks.

It is therefore highly important to develop a system that communicates a range of information
about the environment in a nonvisual manner, to allow a person greater knowledge and enjoyment of, connection to, and more effective navigation through the space. An appropriately developed wayfinding system can enhance our ability to (1) keep track of our current location and heading as we move about, (2) find our way around and through a variety of environments, (3) successfully find and follow an optimally safe walking path to our destination, and (4) be aware of salient features of the environment.

Developing such devices is difficult and time consuming for a number of reasons. First, there is a need to decide what information about the world is relevant to navigation and wayfinding. There has been some limited research in this area [e.g., Blasch, 1991], which can provide guidance. Then there is the need to survey the environment directly around the user to determine which of the necessary elements are present. This may be done in a rudimentary manner using sonar, radar, computer vision, or other obstacle detection systems, but is more effectively accomplished in modern systems by determining the user's location and querying a database of features around that location. This presents a whole other set of hurdles to overcome, including the obvious need to populate the database in advance. Once those two challenges are overcome, there must finally be a way to communicate information about the presence and location of these environmental features to the user. The present article is focused mostly on this third stage in the process, although all are necessarily intertwined.

Auditory displays are a natural choice for displaying information about a blind person's environment and routes, and several have been developed [see Massof, 2003, for a recent review]. Almost all auditory navigation aids have used recorded or synthesized speech to communicate to the user. Such systems can be easy to learn and operate, but there remain concerns about these speech interfaces, such as privacy and comprehensibility in noisy environments. At this point it is also important to note that there is a significant percentage of persons with visual impairments who also have some hearing loss (nearly 50% for seniors, according to Mortensen, 2000), and this will affect the appropriateness of an auditory display for that user population. However, for those who are able to hear, auditory cues can be very useful in providing information that the visual system cannot. In any case, it is important to ensure that all interface designs are tested carefully for effectiveness in communicating the intended information to the user.

The design process is hampered by the fact that there are few tools available to create and evaluate auditory displays of any type, and, as Walker (2002) points out, there are virtually no guidelines for what they should look like in any case. Much of the development time for auditory displays is still spent on important, but slow, iterations of a design prototype, which also requires access to many listeners who can test a display design and provide suitable feedback. Further, the hardware needed to make such an auditory navigation display feasible is currently bulky and expensive, which makes rapid testing difficult and may not be truly representative of interaction with the final (commercial) product. Finally, any organization that might have the infrastructure and participant populations to be able to develop and test such an auditory navigation system will have an Institutional Review Board (IRB) with real and legitimate concerns about the safety of any system meant to guide people along the sidewalks of a city without being thoroughly tested first in a more controlled environment. An increasingly effective and viable alternative is to prototype the auditory display in a virtual environment (VE) and take advantage of the many benefits that approach brings. Design of products ranging from keyboards to airplane cockpits is now typically done in part with a virtual prototype of the interface. When considering the development of navigation and orientation aids, this is the type of VE research that Durlach et al. (2000) categorized as Type 4C, in which VEs are used to assist in the development of real-world performance aids. It is similar to, but clearly distinct from, using VEs to train a person to perform in the real world, such as the research by Dean Inman at the Oregon Research Institute focusing on teaching blind children to cross the road (http://www.orir.org). That would be categorized as Type 4A use of VEs, according to Durlach et al. (2000).

**SWAN AUDIO INTERFACE**

As part of the development of the Georgia Tech System for Wearable Audio Navigation (SWAN; see http://sonify.psych.gatech.edu/research/), we have encountered exactly the hurdles described above and have, as a result, begun to use VEs for much of the interface development in order to keep our primary project on track. We discuss here our very successful use of a VE to help in the design, evaluation, and iterative refinement of the SWAN auditory navigation system. In particular, we
highlight the kinds of research and design ques-
tions about the (real) interface that the virtual ver-
tion allows us to study. In addition, we point out
some of the ongoing challenges that researchers in
this field need to continue to consider when using
a VE to prototype an auditory interface.

The complete SWAN system employs an array of
sensors (e.g., global positioning system (GPS),
InertiaCube, infrared emitters) and some sophis-
ticated sensor fusion methods to determine the
user’s precise location and orientation. A wearable
computer then determines what features in the en-
vIRONMENT need to be presented to the listener
and renders the appropriate sounds via stereo head-
phones. For our present purposes, we focus mostly
on the design features of the actual auditory inter-
face, rather than the details of the hardware sys-
tem. We should mention, however, that the tech-
nical challenges in simply determining the user’s
physical location using commercial tracking tech-
nology are substantial. For example, standard
GPS signals have accuracy that is too poor (about
3 m at best) to be relied on alone. The GPS reading
can also jump around and occasionally give mas-
sively incorrect readings for a second or two. This
is a challenge faced by many systems of this type,
and several researchers, including ourselves, are
tackling it with good success. However, since the
engineering issues are not 100% solved, it is im-
portant to be able to design and study the interface
that the user will experience separately from the
physical hardware system.

The SWAN interface utilizes a repertoire of non-
speech auditory icons and earcons within a specific
framework to allow users to navigate successfully.
The environmental elements that are sonified in
SWAN include navigation waypoints; a variety of
objects like offices, benches, and mailboxes; and
surface transitions such as carpet to tile, or level
corridor to ascending stairs. “Beacon” sounds rep-
resent the location of waypoints along a path and
are used to accomplish the primary wayfinding
 task (see below). The other sounds are used to con-
voy knowledge about the features in the world and
allow exploration of the immediate environment
and enhance situational awareness. Sounds rep-
resenting these features are processed dynamical-
ly and presented through the headphones in such
a way that they seem as if they were located at the
Corresponding real-world location (i.e., they are
spatialized). For example, if the real environment-
feature (e.g., an office door) is ahead and to the
right of the user, the sound that represents the of-
Rice number and office occupants will appear to em-
 Mane from a location in front and to the right of

the user. Note that only the SWAN user hears the
sounds, thus not disturbing other people who may
be around.

A complete path that a user might wish to travel
is broken down into shorter, straight, unobstruct-
ed path segments that are joined by waypoints. In
order to move along a premade path, the user selects
for the beacon of the next waypoint and simply
walks toward its apparent location. Once the user
reaches the waypoint indicated by the beacon, the
sound shifts in the listener’s three-dimensional au-
ditory space to the location of the next waypoint;
the user then reorients to the new beacon location
and selects the next path segment. Thus, a cru-
cial element of the system is the ability of the user
to localize the beacon sounds in the three-dimen-
sional audio space. Since at this point SWAN uses
generalized head-related transfer functions (HRTFs)
to spatialize the sound, the more we can do to
help the listener in this auditory localization
 task, the better.

THE SWAN VE SANDBOX

As discussed, there are many challenges in de-
signing an auditory display, especially a sophisti-
cated one such as that used by SWAN. In partic-
ular, there are many novel design elements, not
least of which is the use of nonspeech audio cues,
which need to be examined critically for their ef-
ectiveness. The VE prototyping environment we
have developed allows us to implement and rapidly
evaluate our sounds, menus, and interaction de-
vices in a safe and controlled laboratory environ-
ment before testing with the full SWAN system.
Our VE was constructed using the Simple Virtual
Environment (SVE v.1.5) software package devel-
oped by the College of Computing at the Georgia
Institute of Technology (GVU Virtual Environ-
ments Group, 1997). SVE is run on a Dell Optiplex
PC running at 1.7 MHz, with 512 MB of RAM. Es-
Sentially, the SVE component provides the rest of
the SWAN system with input that mimics the user
walking around in the world. Rather than actually
logging the full SWAN system and walking around
subject to the localization issues inherent in the
outdoor system, the participant stands in
the center of a quiet laboratory, listening via head-
phones. To simulate walking forward, the user
pulls the trigger of a modified joystick, known as a
flightstick (he or she does not actually walk for-
ward). To change direction, participants rotate
their body on the spot where they are standing as
their orientation within the environment is
tracked by an Inertence InertiaCube 2 head-
mounted tracking cube attached to the head-
phones. The SVE system sends out the user's lo-
cation to the rest of the SWAN system in simulat-
ized GPS coordinates, at which point the SWAN's audio code is used to render the sounds, regardless of whether the user is actually strolling through the campus or navigating virtually in the laboratory. As a result, the auditory interaction, including physically orienting to three-dimensional audio sounds, is identical in the VE environment and full SWAN systems. Other than the movement meth-
ood, our testing with participants to date has re-
vealed that users of both systems do not report any major differences in the experience. Thus, we have a virtual version of the system that is safe, effective, and can be used to rapidly prototype and test sound design concepts.

KEY QUESTIONS

The use of the VE version of the interface allows us to address several important questions in the development of the full SWAN system's audio in-
terface. The flexibility of the VE testbed allows us to be very thorough and exhaustive, testing a much greater number of sounds and configurations than would be possible without the VE. It does not, of course, remove the need to test and verify the sound design while walking around with the real SWAN system, but it does provide a highly useful prototyping tool for certain questions. Since there are so many possible aspects to study, we have begun with the navigation beacons, and we primarily discuss those now. We note that the aim of the present article is to discuss the issues relating to using VEs in designing auditory interfaces. The full details of the separate experiments have been or will be presented elsewhere. In addition, ongoing and planned studies are investigating the inclusion and design of the sounds that represent the other environmental features.

Sound Design

Since sound localization is so crucial for the SWAN, the first question we addressed was what types of sounds would lead to the most effective navigation performance. Based on a number of human factors and perceptual issues, as well as the findings of Tran, Letowski, and Abouchacra (2000), we have decided to use nonspeech beacon sounds (for more discussion on this issue, see Walker & Lindsay, in press, as well as Stokes, Wickett, & Rite, 1988). The question then becomes one of how to design the nonspeech sounds. Clearly, it is useful to make each initial design de-
cisions using the VE. We had listeners navigate through three separate maps in the VE using one of three different beacon sounds. The VE in which these maps were located was essentially a large empty (virtual) room 100 m² in size. No visual fi-
deliness is required, since the participant has no vi-
sual display. This makes the use of a virtual test-
ing environment even more compelling for audi-
tory display design, since relatively little work is required to obtain a suitable simulation. The three maps differed simply in the layout of the waypoints and in overall path length (see Walker & Lindsay, 2003). The three beacon sounds were each 1 second long, with a center frequency of 1000 Hz and equal loudness. The sounds differed greatly in timbre, however. The first sound beacon was a burst of broadband noise centered at 1000 Hz. The second beacon was a pure sine wave with a frequency of 1000 Hz. The third beacon sound was a sonar pulse, similar to the sound that Tran et al. (2000) found to be one of the best sounds for use as a nav-
igation beacon. Each participant navigated using the same sound throughout his or her three maps. At the start of a map the beacon sound played in an on-off pattern, with the sound on for 1 second and off for 1 second. As the listener moved closer to the next waypoint, the silence (i.e., the "off" por-
tion) was shortened to effectively make the beacon tempo faster. Hence increasing proximity to the waypoint was mapped to increasing tempo, which is consistent with our findings for population stereotypes or preferred mappings between proximity and tempo (Walker, 2005).

The results of this study (Walker & Lindsay, 2003) showed a significant difference in performance for the different sounds, such that the broadband noise beacon led to the best perform-
ance overall. This is not surprising, since a key to success in the navigation task is being able to localize the sound in space. The broadband noise should be most localizable based on its multiple spectral cues. However, we also found that the best sound to use depended on the other details of the audio interface (discussed next). In some cases, the noise beacon was not the best performer; rather, a sonar-like ping was more effective. This finding highlighted the complexity of developing a novel interface and pointed out the need for further ex-
periments. Using the VE testing system, we were able to be systematic and scientific and include a full range of capture radius conditions to probe the details of the beacon sound results. We discuss these investigations next.

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Capture Radius

In addition to the sounds used to represent a waypoint, the particular interaction a listener has with the waypoint is important to consider. Each waypoint is specified by exact x, y, and z coordinates. However, the precise location of the user might never exactly "reach" the waypoint's location. Consider the following analogy: A person starts at Point A, walks down the sidewalk to Point B on the corner, turns left, and continues along the sidewalk to Point C. There is a penny on the sidewalk at the corner, indicating the exact place to turn (the waypoint, or Point B). The typical user walks the first path segment, reaches the waypoint, turns the corner, and completes the path successfully. However, she might never actually step right on top of the penny at the waypoint, despite passing pretty much right over it. A computer system might say that she failed to traverse the path correctly, since she never technically arrived at the penny-sized point. A human observer would, on the other hand, say she was definitely "close enough" to each of the points. This points to the need for a capture radius. That is, there must be a radius around the waypoint that is considered close enough, so that the next beacon sound can appear and the user can carry on down the next path segment. If the capture radius is too small, a person might "overshoot" the waypoint, walk past the corner, off the sidewalk, and into the street. If the capture radius is too large, the user may be told she has reached the turning point too soon, and as a result change direction toward the next waypoint and either cut across the grass or run into the corner of a building in doing so. Thus, to keep the person on the intended path—neither missing the marks nor turning too soon—an optimal capture radius needs to be determined.

Rather than deciding on an optimal capture radius simply by trial and error with the real system, we were able to use the VE to go a step beyond and try out a number of capture radii in controlled conditions. We explored small (30 cm), medium (1.5 m), and large (15 m) values and allowed dozens of participants to move through a variety of paths as described above. An example of the results is presented in Figure 1. Note how with the smallest radius (top panel) participants spend a considerable amount of time hunting for the waypoint. This would clearly be unacceptable in the real world. In the large radius condition (bottom panel), listeners often never reached the waypoint and generally deviated from the prescribed path too much for practical navigation success. Participants using

FIG. 1. Navigation performance with three different capture radius conditions when using the sonar beacon. The top panel shows the results from the small radius condition, in which hunting behavior is evident, leading to some backtracking and inefficiencies in path traversal. The bottom panel shows the large radius condition, where considerably more erratic movement is evident. The middle panel shows the medium radius condition, in which the participants' movement paths closely follow the scheduled path, with relatively few oversteps, hunting, or other inefficiencies. The medium capture radius led to the best performance overall.
the medium waypoint (middle panel) showed successful navigation, neither straying too far from the path nor spending much time trying to find the waypoint. The results discussed here are part of a much larger study (see Walker & Lindsay, in press). Without the VE, data collection in such a large study would have meant transcription hours of videotape, estimating scores of movement paths, assessing interrater reliabilities, and so on. The VE testing environment allowed us to gather a considerable amount of precise quantitative data, thereby allowing for a careful statistical analysis of both speed and accuracy. Since navigational accuracy is our primary concern in the development of the SWAN system, we can use these results to determine that the medium capture radius was most appropriate, and based on that decision we can consider that the sonar beacon may, in some cases, be better than the noise beacon. Thus, the scientific method, in conjunction with effective VE tools, allows us to perform iterative investigations to converge on a more sophisticated understanding of user interaction with the system we are developing.

**Practice and Training Effects**

The issue of how the use of the system changes with practice is crucial to understanding the effectiveness and utility of the device. Although practice has not been specifically isolated and investigated yet, our findings to date show large improvements in performance by participants over a relatively small number of trials. Over the course of only three trials, during which the virtual path navigated became progressively more complex and difficult, participants showed a remarkable degree of improvement in both rate and efficiency. In fact, the movement traces shown in Figure 1 represent only the second map for participants. That is, after only a few minutes of experience with the system users are already comfortable and quite proficient, given the constraints of the interface. It is important also to consider that after the initial instructions on how to navigate using the auditory cues in the SWAN system, no more guidance was provided. Hence a small amount of initial instruction coupled with only three trials worth of practice (typically less than a total of half hour of experience with the system) resulted in very large increases in proficiency by the participants. At the conclusion of those trials, performance was still improving, indicating continued practice could lead to even greater gains. Studies currently underway with the VE system are examining the asymptotes of proficiency.

Given the improvements noted in the VE system, it makes sense to use the VE to help train users of the outdoor system. The VE allows the task to be broken down into component parts, such as the analytic listening, sound localization, sound recognition, and movement components. This task-decomposition training has been used effectively in other auditory display training approaches (e.g., Smith, 2005; Smith & Walker, 2000). However, this is a different kind of use for the VE, namely using the VE testbed to train a person to perform in the real world (Type 4A, according to Durlach et al., 2000). This is not specifically what the SWAN VE test system was developed for. All indications with participants and researchers who have used the two versions (virtual and real) lead us to believe that practice in the VE transfers to the outside. However, this remains to be systematically evaluated. In any VE, it is important to at least admit the possible limitations of this Type 4A activity. We are confident of the transfer, but must still exercise caution. Even if the whole task does not completely transfer (although we view this as very unlikely), it is still clear that the specific listening skills gained in the VE (e.g., analytic vs. holistic listening, localization of spatialized sounds, etc.) will transfer since the tasks are identical in the two environments. The veridicality of the auditory display in both situations is crucial to this effective transition between environments. It is considerably more difficult to develop a visual interaction in a VE to the same level of realism as an auditory VE can be developed, but for our auditory-only design that is clearly not a problem.

**Front-Back Confusions**

Despite the use of generally well-localizable sounds, and then within that class of sounds experimentally determining the best sounds, front-back confusions remain a potentially serious problem for any audio navigation interface. That is, listeners sometimes can report that a virtual sound that was supposed to be in front of them sounds more like it is behind them, in the virtual audio space. This typically arises from the slight discrepancies between the listener’s exact individual HRTF and the generalized HRTF employed by the system for practical reasons. For many purposes the generalized HRTF is sufficient, which is good because the effort involved in measuring an individualized function is considerable and can in some cases outweigh the benefit of using the VE for

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testing and development, in any case, it is obvious that a person not knowing whether to move forward or backward is a significant issue, although it is a rare occurrence. In the first version of the SWAN interface, the only methods to resolve front-back confusion involved turning the head and listening for any spectral cues, or moving forward or backward and listening to the tempo changes in the sound. However, our early results suggested that a more formal method of resolving front-back confusion might improve performance. For example, the hunting behavior seen in the top panel of Figure 1 shows that when walking past a waypoint, a certain amount of time passes before the listener determines the overshoot has occurred and turns to correct the error. If such an overshoot could lead to stepping off the sidewalk and into the street, this could be a very serious problem (although it is somewhat rare in the medium capture radius condition). One method currently being examined in the VE testbed to make an overshoot more salient, and to reduce front-back confusion in general, is the use of beacon sounds that change categorically based on whether they are in front of or behind the user. This is just one of many possible ways to reduce front-back confusion, and with the use of the VE it is much easier to quickly test multiple possibilities and determine what is the most useful disambiguation scheme. In recent studies, first with the VE and then in the real SWAN system, we have actually determined, somewhat to our surprise, that the disambiguation sounds tend to harm performance. It turns out that the listeners begin to attend too much to the change in the sound type from “in front” sounds to “behind” sounds, and end up not attending to the location of the sounds themselves, which is a more reliable cue. Without the VE studies, we may have assumed that the disambiguation sounds would help and implemented them directly. Instead, we are now implementing follow-up studies in the VE to determine if other forms of front-back disambiguation methods can be effective. It is important to understand whether it was the concept of disambiguation that was ineffective, or just the specific implementation. We suspect the latter, and the VE will allow us to explore a range of alternative options before implementing any successful approaches in the full SWAN.

Comparison of Sighted and Blind Users

Visually impaired individuals are a key target user population for audio navigation systems such as the SWAN. However, there is prior evidence (Walker & Lane, 2001) that some aspects of auditory interfaces may not be interpreted the same by sighted and visually impaired users. In that study, participants responded to sounds designed to represent data values such as temperature, velocity, or distance. There were many similarities, but also some important differences in the way sighted and visually impaired listeners interpreted the sounds. Hence it is crucial to test auditory interfaces with members of both sighted and blind participants and look for any differences in usage. A study is currently underway at the Atlanta VA hospital examining these questions using the VE. This allows us to safely determine if and where there are differences in the way the interface needs to be designed, depending on the listener. The visually impaired participants are just as able to use the spatialized sounds to navigate, and in fact the most proficient participant we have had so far (across all studies, all groups) is visually impaired. Preliminary results seem to indicate that there are few if any differences in use between sighted and nonsighted listeners overall, but full conclusions will not be able to be made until more data have been collected and analyzed. Further, studying important but more subtle issues such as whether there are any differences between congenitally and adventitiously blind participants will require considerable numbers of participants from a range of backgrounds. Regardless, this step in the design evaluation is necessary and would not be possible without the VE. We can, thankfully, take feedback from the visually impaired participants who have participated and make changes to the system on the fly, if necessary. It is interesting to note that the visually impaired listeners seem to have very little difficulty adapting to the paradigm of moving in a VE that has no visual component. Anecdotal evidence thus far indicates that they might have an easier time adapting to such an interface since members of this group are already used to navigating with no or little visual information and using auditory cues as important sources of information. This has to be balanced, however, with the generally lower level of experience with VEs, computer games, three-dimensional drawings, and so on, as compared with the sighted participants in our studies. Generalizations of the results always need to be very careful in such circumstances, and verification with the full SWAN system will be required, of course.

Human-SWAN Interface: Input

In addition to the auditory-output components of the system, it is crucial that the visually impaired
user be able to input instructions to the system. It is not a trivial matter to develop a successful interface methodology for a system that is both mobile and intended for use without vision. We can approach design decisions from the analytical perspective (e.g., using task analysis or needs analysis) or from the empirical perspective (i.e., employing the results of systematic studies). With the VE, we have the luxury of utilizing both methods.

We might, for instance, decide the question of whether or not to implement speech recognition as a means of commanding and controlling the system. It is a complex decision process involving a variety of opinions that are perhaps split on the issue. Proponents would consider speech to be a natural, hands-free, eyes-free interaction method that is highly appropriate to the task. On the other hand, due to the wear-anywhere design goal, task analysts might suggest that a speech interface (i.e., voice recognition) is not the most appropriate because of privacy concerns, among others. To clarify, if the (visually impaired) user wished to withdraw some cash from the bank, speaking aloud a phrase like, “System, take me to the nearest ATM” while standing on the street corner could have serious personal security issues. Further, understanding the technical limitations of speech recognition technology “in the wild,” we would predict that the issues of background noise, varying sound levels, wind, and so on make the speech recognition sub-system considerably more error-prone, which is unacceptable in an assistive technology of this sort. Thus, a priori we might have theoretically motivated reasons for avoiding speech recognition in this particular design.

We might then attempt to resolve these opposing design recommendations (speech input vs. non-speech input) by testing it in the VE. We could implement a speech recognition interaction system and see how well it performs. We did that, and it can perform (technically) quite effectively with the virtual system in the laboratory. However, in the end we decided against that approach for the real system. The reason for bringing this up is that the use of the VE to develop a system could lead a team to implement something like speech recognition without realizing that the actual usage context may be quite different from the test environment.

Clearly, this discussion shares much with the discussion of transfer of training above. In that case, the recommendation was to make the actual listening task as identical as possible in both practice and live situations. In the present discussion, the key is to do two things: first, conduct a careful task analysis to determine the environmental context and situational factors that will impact on the final system usage and, therefore, the final system design. In general, these considerations need to be given substantial weight in the decision process.

Second, make sure that additional “environmental” sounds are present in the VE when appropriate. For example, we are able to add sounds in the virtual world that move and change as a real sound source would. The VE allows us to arbitrarily place many complex sounds into the world so that they become audible when the listener comes close enough to the source, or when some other condition is satisfied (including randomly timed sound occurrences, like virtual birds chirping). The VE handles all of the associated calculations, attenuations, interaural difference cues, and so on. This approach preserves scientific control while introducing a measurable measure of ecological validity. This is in keeping with the growing trend toward more ecological psychosocial research in the development of auditory displays (see Walker & Kramer, 2004). In addition to adding sounds in the VE itself, additional sounds may need to be added to the testing room external to the VE. For example, recorded traffic noise can be played over loud speakers surrounding the participant. The reason for these external sounds is that the internal sounds in the VE, especially if played over headphones, will not impact on the speech detection capabilities of the system. The addition of external sounds within a testing room is an important example of where the VE is limited and needs to be supplemented in order to maintain careful evaluation of the input capabilities of the system.

In addition to the task-analytical methods already discussed, these “noise supplement” studies with the VE are the sorts of assessments that have led us to use manual (button-press) methods for input, and to use speech messages (presented privately to the user) only for output. Audio menus built in this manner have yielded effective and, perhaps most importantly, user-acceptable interaction with the SWAN system. The menu options are presented visually. The user selects the desired functionality from the menu using buttons on a small wireless handheld device kept in a pocket or attached to a belt loop. The key here is that the need for such a system came about primarily through testing alternative input methods (speech) in an enhanced VE that incorporates environmental noises, and partially through task-analytical approaches that led to conclusions that the VE alone would have never produced. The virtual system alone may not have pointed to the need for privacy or the difficulties of speech recognition in the open usage environment. **Auditory Displays and Virtual Prototypes**

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DISCUSSION

There is a clear need for assistive technology to help persons with vision loss continue to navigate and learn about the environment (Massof, 2003; see also www.ibis.org). The use of a VE can be extremely useful in developing, evaluating, and iteratively improving technological solutions, such as the SWAN being developed at Georgia Tech. The key is that the VE enables rapid, safe, and systematic studies in ways that are simply impossible in a trial-and-error design approach. Examples of the benefits include sound design for maximum localizability; details of how the system interoperates with the user (e.g., capture radius); and an examination of practice effects. When novel or unexpected findings emerge, such as the interaction of capture radius and beacon sound in our system, the flexibility of the VE and its utility for gathering data enable researchers to probe the details of how the system performs.

It is crucial to recognize, however, that becoming entirely reliant on the VE can lead to design flaws and potential safety issues. We discussed here how the exclusive use of the simulator might lead a designer to implement, for example, a speech recognition system simply because it is relatively simple to do in the virtual system. It should be stressed that the full consideration of task needs must be included in the design process. That is, the VE must never rise above the status of being just one of the many tools employed by the assistive technology developer.

On a related note, it cannot be over-emphasized that although there are many similarities between indoor VE test systems and outdoor full-use devices, ultimate validation studies definitely have to be part of the development plan. Further, whatever possible an ecologically valid testing environment needs to be utilized. Having said that, however, it is also important to recognize that displays using certain modalities (notably auditory display) are generally easier to duplicate exactly, or nearly so, in a VE test system. For this reason, with careful considerations always in mind, developers of auditory assistive technologies may feel quite comfortable in making the cost of what is proving to be a very effective tool to enhance the science in assistive technology design. VEs, used in conjunction with more traditional design methods such as task analysis and user-centered design, play an important role in ensuring not only that assistive technology is available, but that it is also effective, usable, and acceptable.

REFERENCES


Smith, D. R. (2003). Effecting training and learning in a point estimation navigation task. Un-

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published master's thesis, Georgia Institute of Technology, Atlanta.


