

3D Localization and Tracking of Objects Using Miniature Microphones

Radu Ionescu¹, Riccardo Carotenuto^{1*}, Fabio Urbani²

¹DIMET, Università "Mediterranea" di Reggio Calabria, Reggio Calabria, Italy ²Department of Engineering, the University of Texas at Brownsville, Brownsville, USA E-mail: r.carotenuto@unirc.it Received March 1, 2011; revised April 1, 2011; accepted April 11, 2011

Abstract

A system for accurate localization and tracking of remote objects is introduced, which employs a reference frame of four coplanar ultrasound sources as transmitters and miniature microphones that equip the remote objects as receivers. The transmitters are forced to emit pulses in the 17 - 40 kHz band. A central processing unit, knowing the positions of the transmitters and the time of flight of the ultrasound signals until they reach the microphones, computes the positions of the microphones, identifying and discarding possible false signals due to echoes and environmental noise. Once the microphones are localized, the position of the object is computed by finding the placement of the geometrical reconstructed object that fits best with the calculated microphones positions. The operating principle of the localization system is based on successive frames. The data are processed in parallel for all the microphones that equip the remote objects, leading to a high repetition rate of localization frames. In the proposed prototype, all the computation, including signal filtering, time of flight detection, localization and results display, is carried out about 25 times per second on a notebook PC.

Keywords: Localization System, Remote Object, Tracking, Ultrasounds, Time of Flight

1. Introduction

The increasing interest in systems able to provide users with remotely accessible capabilities (e.g. security, domotics, health care, new generation of game consoles and video games, etc.) has encouraged the development of cheap and effective devices aimed at tracking objects and people within a certain space region. Accurate objects localization and tracking is currently a challenging problem. Tracking is normally performed in the context of higher-level applications that require the location and/or shape of the object at every iteration or time instant. Difficulties in tracking objects can arise due to object masking by external obstacles or abrupt object motion.

A visual approach using video cameras is employed in most applications, when objects localization and tracking is based either on color hystograms [1-3], illumination changes [4], occlusion [5-7], appearance [7,8] or scale variations [9]. Infrared techniques can also be applied [10-12]. An important drawback of video localization systems is that they cannot be used in many situations or

environments due to the frequent blockage of the light by different obstacles and structures. Moreover, video localization relies on the camera resolution, generally resulting in poor spatial resolution.

A variety of techniques have been developed for localization purpose, which are based on radio frequency (RF) [13,14]. Probably the most famous one is the Global Positioning System (GPS), but it does not provide a sufficient resolution (order of meters) for some applications such as precise localization of objects and persons, and it is not effective in most indoor environments or other areas with limited view of the sky. Another technique, Global System for Mobile communications (GSM), showed an uncertainty of tens to hundreds of meters in objects localization [15,16]. By employing 29 different GSM channels, a median accuracy ranging from 1.94 to 4.07 m was obtained in indoor objects localization [17]. The use of Wireless Network Technologies, such as Wi-Fi [18], Bluetooth [19], Wireless Local Area Networks (WLAN) [20,21] or ZigBee [22], did not achieve better localization accuracy. Ultra-Wide Band (UWB), Indoor GPS positioning and Radio Frequency

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Identification (RFID) technology were also evaluated, providing an uncertainty in estimating mobile objects localization in the order of cm [20,23].

The sound source localization is based on determining the coordinates of sound sources in relation to a point in space. In a recent paper, sound source localization and tracking method using an array of eight microphones was proposed [24]. The method is based on a frequency-domain implementation of a steered beamformer along with a particle filter-based tracking algorithm. Using an array of 8 microphones, a robot was able to follow a person speaking to it by estimating the direction where the speech sound was coming from. The localization accuracy was around 1° within 1 m distance, both on azimuth and elevation, and around 1.4° within 3 m distance. The current location of another robot, using an array of 24 microphones distributed on two walls inside a close laboratory room, was based on robot speaking, and produced an average localization error of about 7 cm close to the array and 30 cm far away from the array [25]. These are examples of active localization systems, in which the reference system is equipped with receivers placed at known locations, which estimate the distance to the remote device based on acoustic signals transmitted from the device.

A similar strategy was employed in the case of the active Bat ultrasonic location system for people localization [26]. Small units called Bats, consisting of a radio transceiver, controlling logic and ultrasound transducer, are carried by persons. Ultrasound receiver units are placed at known positions on the ceiling of an indoor room. The times-of-arrival (TOA) of ultrasound from the Bat emitting device to each transducer are measured, and radio signals are used for synchronization. The location accuracy was below 10 cm.

The 2D position of an automatic guided vehicle was obtained with an accuracy of a few mm from the time-of-flight (TOF) of ultrasound signals [27]. The localization system employed consisted of ultrasound reception and emission beacons positioned at the same height on the docking workstation and on the automatic guided vehicle, respectively.

The Massachusetts Institute of Technology has developed the 'Cricket' indoor location system. 'Cricket' uses a combination of radio frequency (RF) and ultrasound signals to obtain the location of a remote device. Beacons placed on the walls and ceilings inside a building transmit a concurrent ultrasonic pulse on each RF advertisement. When this pulse arrives to listeners attached to the remote device, these estimate the distance to the corresponding beacon by taking advantage of the difference in propagation speeds between RF and ultrasound. This method employs a passive localization system, in which

the reference system is equipped with beacons placed at known locations that periodically transmit signals to the remote device equipped with receivers, which estimate the distances to the beacons. The Cricket beacons and listeners are identical hardware devices. The Cricket system could provide positioning precision between 1 and 3 cm [28].

Most 3D-localization systems based on ultrasound distance measurement use time-of-flight measurements which can be easily and cost-efficiently performed because of the slow speed of ultrasound in air (about 343 m/s at 20°C). These systems include either a few reference beacons (minimum 3 or 4) equipped with ultrasound transmitters to localize receiving devices, or vice versa, the localized device transmits an ultrasound signal received by several microphones belonging to a reference systems. Transmitted ultrasound signals are realized as constant-frequency bursts or coded signals in a broader frequency band [29]. The principal advantages of a localization system with transmitters in fixed locations and receiving sensor devices are that the device is able to compute its own position locally, and that the transmitters can send signals synchronously [30,31].

Recently, we have presented promising preliminary results for very accurate objects localization and tracking, employing a new approach based on a passive localization system [32]. In this paper we show the capabilities and achievements of our system. It is quite different from the Cricket localization system, which is composed of complex and intelligent nodes that allow easy cellular management. The latter is useful especially in a multi-room environment, but it shows relatively low positioning accuracy and rate. Moreover, the single node results quite big and not easily worn or placed on small objects. No applications employing the Cricket system were reported on gesture tracking and fine positioning, which in fact require localization rates in the order of tens of times per second and accuracy in the sub-centimetric range. The localization system that we propose is much simpler both in construction and operation mode and, in perspective, well suited for future system-on-chip realizations.

The proposed system employs a reference frame of ultrasound sources as transmitters, and miniature microphones that equip the remote object as receivers. The distances between transmitters and receivers are estimated from the time-of-flight of the ultrasound signals. Ultrasound based time-of-flight methodology was proved to be more reliable and accurate than radio based approaches [33]. Moreover, the use of ultrasound sources and miniature microphones reduce at a minimum both the audible and dimensional discomfort during system operation.

The paper is organized as follows: in the next section we describe the proposed system and its principle of operation, while in the following sections we present the realized prototype, the localization algorithm employed and the experimental results obtained in different applications.

2. System Description and Principle of Operation

The aim of the system that we developed is to obtain the accurate localization and tracking of the movement of remote objects. The main elements of the system are a set of ultrasound sources and one or more receiving targets. In our localization system, we employ tweeters as emitters, being the active elements of the system, and miniature microphones as receivers, which are the passive elements of the system.

The reference frame of the system is formed by four tweeters placed at known locations, specifically at the vertexes of a rectangle with the lengths of its sides *a* and *b*, respectively (**Figure 1**).

The remote object is equipped with several miniature microphones mounted at strategic positions on the object, which are wired to a data acquisition board. Knowing the coordinates of the microphones and the geometry of the object, the shape of the object and its 3D orientation can be represented in any virtual context. The strategy that we propose is limited to devices that do not deform significantly their shape when they are manipulated.

The operating principle of the localization system is based on successive frames. For locating the current position, or tracking the movement of the remote object, the localization of each individual microphone at every

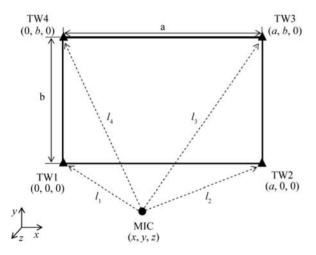


Figure 1. The reference frame with four tweeters (represented by triangles) and one microphone (represented by a circle).

given time instants (*i.e.*, localization frame) during system operation is necessary. Once determined the localization of the microphones, the position of the object is computed by finding the placement of the geometrical reconstructed object that better fits with the calculated microphones positions.

The positions of the microphones are calculated in relation with the origin point of the reference frame, which for convenience was chosen to be the position of one of the four tweeters.

During each localization frame, the tweeters emit, at predefined constant time intervals, ultrasonic pulses towards the remote object. These pulses, reaching the miniature microphones placed on the remote object, are acquired as electrical signals by the data acquisition board.

The distance l between each microphone and each one of the tweeters is indirectly estimated by a central processing unit from the time of flight T_f employed by the ultrasound signal emitted by the tweeter to reach the microphone, assuming the linearity of the wave propagation path and that the speed of sound v in a given transmission medium is constant and its value is known (Equation 1). An inherent delay introduced by the system components during signal propagation must be also taken into account. Thus, a constant offset distance l_{offset} , whose value was experimentally measured, must be subtracted in order to obtain the correct distance between each pair microphone/tweeter:

$$l = v \cdot T_f - l_{offset} \tag{1}$$

The obtained distances are then used to calculate microphones positions. For each microphone, the following four sphere equations that describe the distances between the microphone and the four tweeters forming the reference frame can be written (Equation (2)):

$$\begin{cases} l_1^2 = x^2 + y^2 + z^2 \\ l_2^2 = (x - a)^2 + y^2 + z^2 \\ l_3^2 = (x - a)^2 + (y - b)^2 + z^2 \\ l_4^2 = x^2 + (y - b)^2 + z^2 \end{cases}$$
(2)

Resolving the four possible systems resulting from picking in all combinations only three equations at a time from the available four sphere equations, four values for the position of the microphone are determined: (x_1, y_1, z_1) , (x_2, y_2, z_2) , (x_3, y_3, z_3) and (x_4, y_4, z_4) , respectively. Here, we should note that every system of three equations of (2) has two possible solutions, and a single solution is obtained by limiting to a half-space the valid region of operation for the remote device with respect to the reference frame.

(4)

If the estimation of the distances between the microphone and the four tweeters would be perfect, four identical values for microphone position would be obtained. However, due to different disturbances (noise, echoes, obstacles, etc.), l_1 , l_2 , l_3 , l_4 are generally affected by slight uncertainties. By computing the position of the given microphone as the mean value of the four positions calculated on each one of the three axes (Equation 3), any small errors occurred in calculating the position of the microphone are minimized. This represents an advantage provided the fact that we employ a reference frame formed by four tweeters, and not by only three tweeters that would have been otherwise sufficient for estimating the position of the microphone.

$$\begin{cases} x = \frac{1}{4} \sum_{i=1}^{4} x_i \\ y = \frac{1}{4} \sum_{i=1}^{4} y_i \\ z = \frac{1}{4} \sum_{i=1}^{4} z_i \end{cases}$$
 (3)

The robustness of the algorithm developed relies on a technique to discard any fake points, which could imply the failure of the correct localization of the object. For this purpose, the "distance" between all the four computed microphone positions is calculated, i.e. the square root of the sum of squares between every pair of two from the four computed microphone coordinates on each one of the three Cartesian axes (Equation 4):

$$Test = \frac{1}{12} \sqrt{\sum_{\substack{i,j=1\\i < j}}^{4} \left(x_i - x_j\right)^2 + \sum_{\substack{i,j=1\\i < j}}^{4} \left(y_i - y_j\right)^2 + \sum_{\substack{i,j=1\\i < j}}^{4} \left(z_i - z_j\right)^2}$$

If the parameter *Test* is higher than a set threshold value, the localization of the given microphone is con-

However, the object can be super-described by employing a higher number of miniature microphones than the strictly necessary one to represent its shape, e.g. 4 or more microphones for a parallelepiped. By doing this, any failure or error in the correct localization of one or more microphones will not compromise the accurate computation of the position of the object, provided that valid signals from the strictly necessary number of microphones are still obtained.

When the number of correctly localized microphones during a given localization frame is not enough in order to represent the shape of the object, the visualization of the object is skipped during that particular localization frame. However, because of the high repetition rate of localization frames achieved by the proposed localization

system (Section 4), this rare undesirable event could easily pass practically unnoticed even concerning applications in which the object describes a fast movement.

3. Experimental Set-up

The prototype of the proposed system is presented in **Figure 2**. The different components forming the prototype are listed below:

- Processing unit: a PC is employed as central processing unit of the localization system. It uses algorithms written in Matlab (The MathWorksTM) both for building the acoustic pulses that are emitted by the tweeters, and for acquiring, storing and analyzing the signals received by the microphones.
- Data emission/acquisition board: MOTU 828 mk3 (MOTU, Cambridge, Massachusetts, USA). It is provided with ten analog inputs and outputs that can operate at sample rates up to 192 kSamples/s. The PC connection is realized via FireWire.
- Tweeters: Sony MDR-EX33LKP (Sony Corporation, Japan). Preliminary tests performed have shown that this specific model is able to emit sufficiently accurate acoustic pulses in the chosen ultra-acoustic band. It fulfils furthermore our requirements regarding small size, while the diameter of its pressure output hole of only 3 mm ensures a wide emission lobe, well covering the space region of interest at around 20 40 kHz.
- Miniature microphones: FG-6163 (Knowles Acoustics, Itasca, Illinois, USA) is a condenser microphone of cylindrical shape, 2.6 mm length and diameter, 0.79 mm acoustical receiver window diameter, and 80 mg weight. The choice of a very small microphone comes as a need of our application. In the proposed localization system, the microphones are placed on the object whose localization is desired. Thus, they must not represent a discomfort,

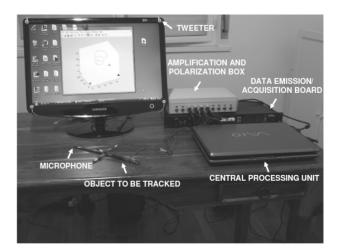


Figure 2. Photo of the localization system prototype.

sidered erroneous.

especially in the case of people localization. Moreover, the small acoustical input window, in respect of the wavelength, ensures a good approximation of the point-like receiver

- Amplification and microphones polarization box. It hosts a power amplifier board and a polarization board:
- A power amplifier board was designed and realized with the purpose to amplify the output signals transmitted by the data board module up to the adequate values able to drive the tweeters. It is provided with four independent channels able to drive simultaneously the four tweeters forming the reference frame of the system with voltage pulses up to 30 Vpp.
- A polarization board was designed and realized with the purpose to provide the polarization of the microphones necessary for their operation.

The shape of the acoustic signals emitted by the tweeters is very important for the localization system. It must be chosen in such a way to be easily identifiable in the electrical signal received from the microphones among other type of possible disturbances contained in the acquired signal (such as acoustical or electromagnetic noises). Furthermore, it must have a well limited bandwidth in order to be well filtered after reception for eliminating out-of-band disturbances.

The acoustic pulse emitted by the tweeters was chosen in the near ultrasound band, which was preferred because many off-the-shelf sound products, in particular tweeters and microphones, can still work sufficiently well for our purposes around and slightly beyond the high corner of their bandwidth. Using ready and mature technologies is a very important aspect for a system intended to become a widely used human-machine interface with very low mass-production costs. Furthermore, signals emitted in this frequency range are non-audible to humans and ensure the acoustic comfort of operation. However, they can actually disturb certain kinds of animals, and this is an issue to be solved in next prototypes using higher frequencies.

The acoustic pulse signal was built using Matlab. It is derived from the discrete anti-transform of a rectangular signal with an open window corresponding to the frequency band set for the emitted acoustic signal (17 - 40 kHz). Actually, for constructing the signal that is finally emitted by the tweeters during system operation, only a few samples (*i.e.*, 120 samples) around the central part of the whole signal were selected, while the rest of the signal was ignored because its power with respect to the noise floor is negligible.

The emitted signal is sent at a sampling frequency of 192 kSamples/s and can be visualized in **Figure 3**. Its central part shows the presence of a unique highest peak and of two equal lowest peaks, which are easily identifi-

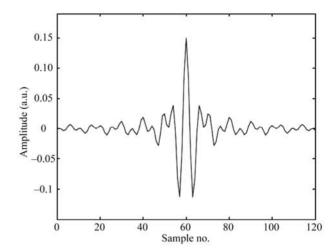


Figure 3. Acoustic pulse signal emitted by one tweeter.

able in the electrical signals acquired. A necessary compromise was made when the total number of samples forming the acoustic pulse emitted by the tweeters, i.e. the truncation level, was selected: its length has to be as short as possible in order to speed up the velocity of system operation, but at the same time sufficiently long for an ease peak recognition in the received signal and for preserving the previously set bandwidth limitation.

All the computation performed by the localization system during each localization frame, starting from the mathematically synthesis of the acoustical signals emitted by the tweeters and ending with the object display, must run with a high repetition rate (>20 Hz) in order to achieve a sort of "real-time" localization and tracking of the object movement. This is obtained by optimizing all process parameters:

- The sampling rate of both data signals emission and acquisition was set to the maximum sampling rate provided by the data board module employed (*i.e.*, 192 kSamples/s). In this way, the duration of the acoustic pulse emitted by every tweeter is of only 0.625 ms (120 samples@192 kSamples/s, **Figure 3**). On the other hand, this high operating rate is also important in order to obtain very accurate object localization, allowing the acquisition of highly accurate and well-shaped signals.
- In order to minimize the communication overhead between PC and data board, a unique sequence is used during every localization frame of the system for outputting the acoustic pulses emitted by the four tweeters. Four different acoustic signals are constructed in Matlab and they are sent simultaneously to the four tweeters through the MOTU board, as shown in **Figure 4**. The peaks of these signals occupy a different temporal position, such that each tweeter emits its signal by its turn.
- A time listening window for data acquisition was determined by setting a maximum distance allowed for

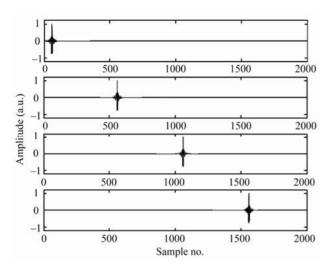


Figure 4. Sequence of the acoustic signals emitted by the four tweeters forming the reference frame during a given localization frame of the system.

the object movement with respect to the reference frame. In particular, the maximum allowed distance for the object movement was defined as being about 50 cm from each tweeter, for which a time listening window of 20.4 ms was set (4000 samples acquired at 192 kSamples/s). In practice, in order to allow a bigger action radius for the object movement and to cover the whole space region, different constellations of reference frames can be placed at strategic positions. A suitable algorithm, which is currently under investigation, must be applied in order to determine from which constellation actually proceed the signals received by the microphones.

- The time interval passing between the acoustic signals emitted by two consecutive tweeters was minimized as much as possible. An important constrain that had to be taken into account was to avoid the overlapping of different signals that proceed from different tweeters impinging simultaneously on the same miniature microphone, because it would make impossible the task of accurately determining the four times-of-flight corresponding to the acoustic signals emitted by the four tweeters. In the case of our application (*i.e.*, object movement up to 50 cm from each tweeter), the minimum duration of this interval was found to be 2.6 ms (500 samples@192 kSamples/s).
- The data processing is conducted in parallel for all the microphones that equip the remote object. Furthermore, signals emission, data acquisition and data processing are conducted in parallel using the Matlab environment. Thus, while a new signal pulse sequence is emitted to the tweeters during a given localization frame, the object location during the previous localization frame is computed by the software algorithm.

4. Results

4.1. Data Processing

Figure 5, up, shows the data signal acquired by the data board module from one microphone during a given localization frame, in which the ultrasound pulses emitted by the four tweeters can be observed.

Because the audio interface connected to the computer uses a software driver to get signals in and out of the computer, there is an inherent latency delay in signals emission/reception by the computer. The value of this delay depends on the recording software, and in our case it was found to have a constant value of about 9.5 ms (1828 samples@192 kSamples/s sampling rate). During signals processing, the developed software algorithm compensates the acquired signals so that they line up with the playback signals.

The first step of the data processing process consists in filtering the acquired signal in order to eliminate out-of-band disturbances and to keep only the useful information contained in the frequency band of the acoustic pulse emitted by the tweeters. In order to perform the filtering step, at first the data signal was transformed from the time domain to the frequency domain by applying the Fast Fourier Transform (FFT), and then a rectangular filter window that corresponds to the frequency band of the acoustic pulse signal emitted by the tweeters (17 - 40 kHz) was applied (**Figure 6**). A high amplitude peak around 45 kHz can be observed in **Figure 6**. It corresponds to the disturbances produced by

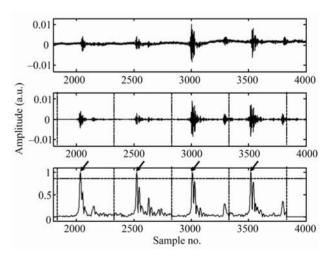


Figure 5. Data signal captured by one microphone during a given localization frame (up); Filtered signal with four defined windows corresponding to the signals emitted by the four tweeters (middle); Normalized values of the absolute value of the Hilbert transform computed for each sequence window (down).

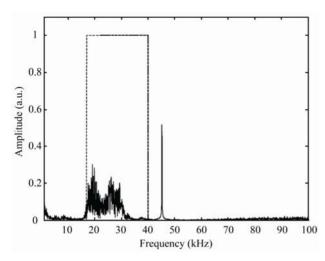


Figure 6. FFT transform of the acquired signal (continue line); Rectangular filtering window (dashed line).

the power supply source, and its presence in the signals acquired by the data board module is sufficiently reduced after applying signal filtering. The filtering process is completed by transforming back the signal obtained from the frequency domain to the time domain.

The next step of data processing consists in determining the time-of-flight of the acoustic pulses emitted by the four tweeters towards every individual miniature microphone equipping the object to be tracked. To this purpose, four equal sequence windows are defined, each one formed of 500 samples that correspond to the time interval between the acoustic signals emitted by two consecutive tweeters (**Figure 5**, middle). For calculating very accurately the four times-of-flight, the absolute value of the Hilbert transform of the filtered signal is computed, which is then normalized for each one of the four sequence windows (**Figure 5**, down).

The presence of echoes or environmental noise in the acquired signal is very likely to produce peaks of considerable intensity. In order to avoid recognizing a wrong signal, a threshold value (set to 0.8, **Figure 5**, down) is defined, and for each sequence window the position of the first sample having an amplitude superior to the threshold value is selected.

The distances between every pair microphone/tweeter are finally calculated using Equation 1. The offset distance l_{offset} was experimentally measured, and it was found to be approximately 2.4 cm. In an indoor environment, the speed of sound in air depends significantly only on the environmental temperature, while the atmospheric pressure is negligible [34]:

$$c_{air} = 331.5 \cdot \sqrt{1 + \frac{T}{273.15}}$$
 (m/s) (5)

where c_{air} is the speed of sound in air and T is the envi-

ronmental temperature expressed in °C.

Once calculated the four distances between a given microphone and the four tweeters forming the reference system, the position of the microphone is computed, as explained above, using Equations 2 and 3. For testing if the localization of the microphone is correct, Equation 4 is applied, in which the threshold value was set to 4 mm that we determined to be small enough to fulfil our requirements in terms of an accurate localization of the microphone. If the number of correctly localized microphones during the given localization frame is enough in order to represent the shape of the object, the data processing during the respective localization frame ends with displaying the object on the computer screen.

4.2. Experimental Applications

The accuracy of the microphones localization was investigated at first. For performing this analysis, a single microphone was placed at different angular positions and distances with respect to the reference frame formed by the four tweeters. The microphone was maintained still in each one of these positions for 1000 localization frames of the operating system. The accuracy of the microphone coordinates localization on the three Cartesian axes over the 1000 localization frames were found to be always below 2 mm.

Different experiments were next performed in order to demonstrate the robustness of the localization system developed for the real-time localization and tracking of objects movement. These experiments are presented in the following paragraphs. For a better visualization, the real-time localization and/or tracking of the object movement was graphically plotted.

a) "Sword" equipped with three miniature microphones.

The first experiment performed had as objective to represent the real-time movement of a sword-like object. For doing this, an ad hoc structure was built, and four microphones were mounted in suitable places on the structure (**Figure 7**, left). It is important to note that the geometrical reconstruction of the virtual object (i.e., the virtual sword) did not intend to reproduce the real shape of the built structure. Instead, the length of the virtual sword, as well as the dimensions of its hilt, were defined in the software algorithm as arbitrary variables whose values can be freely set. In **Figure 7**, it can be appreciated a comparison between the real structure built and the geometrical structure of the virtual sword computed and displayed by the software.

The sword-like object has 6 degrees of freedom, so that knowing the positions of only three out of the four microphones employed to describe the object, the shape

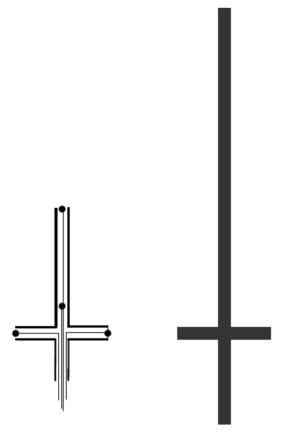


Figure 7. Realized structure of the sword-like object equipped with four miniature microphones represented by black circles (left); Virtual sword geometrically constructed from microphones positions (right).

and the current position of the virtual sword in the space region can be computed by applying a 3D extrapolation. However, the object was super-defined by using a higher number of microphones than the strictly necessary one for defining it. By doing this, we avoid that the erroneous localization of one of the microphones, due to the undesired obstruction of microphone surface by the person handling the real structure, would compromise the correct visualisation during some frames of the real-time sword movement.

The graphical representation of the virtual sword computed at a given time instant during the experiment performed can be visualized in **Figure 8** from different view angles.

This experiment has an obvious practical application in the field of the video-game consoles. The shape of other virtual rigid body object can be geometrically constructed using either the structure that we built, or building different ones.

b) Glove equipped with ten microphones.

The visualisation of the real-time movement of a human hand was the challenging goal of the next experi-

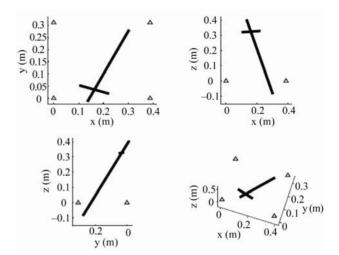


Figure 8. Experimental representation of the current position of the virtual sword from different view angles.

ment performed. A special glove was realized for this purpose using two cotton gloves and ten miniature microphones. The microphones were inserted between two superimposed gloves. The two gloves were chosen very soft and thin, such that not to represent an unpleasant discomfort for the person wearing them. Nine microphones were glued in strategic positions as indicated in Figure 9: they were placed in positions corresponding to the five fingers of the hand: five of them ahead the tips of the fingers, and other four on the finger bones of the thumb, index, middle and ring fingers, respectively. No microphone was placed on the finger bone of the little finger only because of the restriction of the total numbers of microphones signals that can be simultaneously acquired by the data acquisition board employed (10 in total). Instead, for a better intuitive representation of the shape of the human hand, the placement of the tenth microphone at the base of the palm of the hand was preferred. The dashed lines in Figure 9 were plotted for an

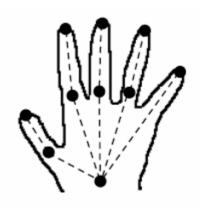


Figure 9. Glove equipped with ten miniature microphones represented by black circles.

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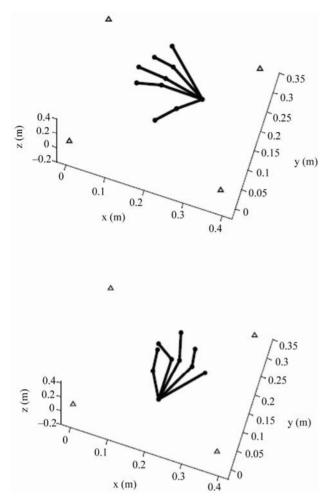


Figure 10. Experimental representation of two different positions of the hand.

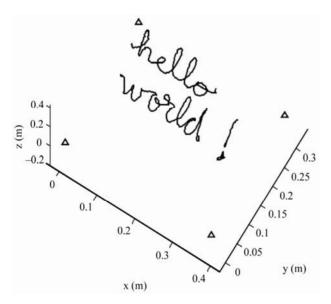


Figure 11. Freehand "air" hand-writing using the virtual pen.

indicative visualization purpose only.

The glove is worn so that the miniature microphones are situated in the part corresponding to the palm of the hand. In absence of any degree of redundancy, the palm of the hand must be held facing the reference frame in order to avoid obstructing the path of the acoustic signals emitted by the tweeters in their way to the microphones. Anyway, this does not constitute any inconvenience in practice, because it represents the comfortable position of the hand during natural movements of man-machine interaction.

The graphical representation of two different positions of the hand during the experiment performed can be seen in **Figure 10**. Although it was out of the aim of our experiment, the geometrical reconstruction of the hand could allow for a better intuitive visualization of the shape of the hand.

As different movements and positions of the hand can be captured using this glove, this could find remarkable applications in the field of domotics and in any field where the natural manipulation of virtual 3D objects is required.

c) Hand-writing.

The last experiment performed was aimed to show the accuracy in trajectory tracking achieved by the system realized. In particular, it demonstrates the possibility to realize a real-time and coherent handwriting in air by means of freehand trajectories described by the miniature microphones.

A virtual pen was thus described by means of two miniature microphones placed on the tip of the thumb and index fingers of the hand, respectively. The writing tip of the virtual pen was computed as the mean distance between the two microphones. During the experiment, if the distance between the two microphones is made shorter than a threshold value (here set to 1.5 cm), *i.e.* the two finger tips approach as they are holding a pen, the virtual pen "writes", while if this distance is made longer than the threshold value, the virtual pen "skips writing". Normal word sequences can be written just playing with the distance between the tips of thumb and index fingers.

The experimental freehand "air" handwriting of "hello world!" using the virtual pen is shown in **Figure 11**.

5. Conclusions

A localization system based on airborne ultrasounds capable of localizing several position markers with sub-centimeter accuracy at a rate of about 25 Hz using off-the-shelf audio components was designed, realized and characterised. The accuracy obtained by our system was below 2 mm in all the three spatial directions within a range of about 50 cm.

Different experiments performed showed that the localization system that we developed allows the real time localization and tracking of the movement of any remote object that does not deform significantly its shape when it is externally manipulated. The proposed system showed a positioning and trajectory tracking accuracy good enough to make it possible a straightforward realization of a gestural interface, which is currently under investigation. At the best of the authors' knowledge, in literature there are no similar localization systems, concerning localization rate and position accuracy. Very promising applications of the localization method here proposed are in the field of gestural interfaces, limb and body movement tracking for medical applications or video game consoles, just to name a few.

Further work will consist in the implementation of wireless communication for the microphones. Higher frequencies, non-audible for the whole variety of living beings, will be used in the next prototypes employing custom transducers.

6. Acknowledgements

R. Ionescu gratefully acknowledges a postdoctoral fellowship funded by the European Commission under the Marie Curie Transfer of Knowledge (TOK) Program (contract no. MTKD-CT-2006-042269). The authors gratefully acknowledge the support of Pentasonics S.r.l., Rome, Italy.

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