

# Scheme Reflections aimed at Positioning Submerged Radar Networks

N.Senthamizh Rajan<sup>1\*</sup> and R.Mala<sup>2</sup>

<sup>1\*,2</sup> PG and Research Department of Computer Science, Maruthupandiyar College, Thanjavur.

[www.ijcseonline.org](http://www.ijcseonline.org)

Received: Aug/22/2015

Revised: Aug/30/2015

Accepted: Sep/24/2015

Published: Sep/30/2015

**Abstract**— The reliance of submerged network deployments on specialized hardware for acoustic communication and modulation has impeded wide scale deployments of submerged sensor networks. This paper examines a system that integrates software modems with generic speakers and microphones that are built-in to sensor modules for establishing submerged acoustic links. Building on our recent field experiment in a river, canal, pond, and swimming pool, this paper outlines the technical and logistical challenges for deploying software-driven submerged sensor network. The software modem design choices include methods for signal modulation at the sender, and symbol synchronization, signal filtering, and signal demodulation at the receiver. Higher layer communication protocol issues are also discussed, with a focus on cross-layer optimization. Practical solutions to logistical deployment challenges, such as waterproofing and casing, calibration, and fouling are proposed. The design guidelines in this paper lay the groundwork for further development of software-driven of submerged sensor networks.

**Keywords**—Submerged Sensor Network, Physical Layer, Synchronization

## I. INTRODUCTION

The design of submerged communication systems has so far relied on expensive specialized hardware for acoustic communication and modulation. The conventional reliance on hardware acoustic modulation has stemmed from low processing speeds that did not allow the modulation of acoustic signals in software.

Software modulation and demodulation is a recent alternative approach which overcomes most of the drawbacks of hardware modems. Recent advances in miniaturization and circuit integration have yielded smaller and more powerful processors that are capable of efficiently running acoustic modulation and demodulation software. Software modulation also provides a higher level of flexibility for on-the-fly tuning of modulation parameters, such as the data transfer rate and symbol duration, to suit the variable conditions of a particular deployment environment.

Coupling software modems with generic microphones and speakers that are built-in to off-the-shelf sensor modules eliminates the need for specialized communication hardware and reduces system cost, facilitating the dense deployment of motes to form submerged acoustic sensor networks. The use of low power generic acoustic hardware and software modems for submerged communication yields a low bit rate in the order of tens of bits per second. The achievable bit rate of the proposed system is sufficient for monitoring sensor networks, such as for environmental or habitat monitoring. In monitoring networks, the nodes sample their sensors and send the data once during each update period, typically in the order of minutes. Since each

node must send only a handful of sensor values during each update period, a data transfer rate in the order of tens of bits per second provides more than enough throughput to communicate all the sensor values during an update period.

The target network application consists of general purpose sensor modules that use software modems and generic hardware to communicate acoustically in shallow water and send the data to the base station, as shown in Figure 1. We expect to deploy a network consisting of tens to hundreds of sensor modules in a shallow water environment. The sensor modules can communicate acoustically through wireless multihop links.

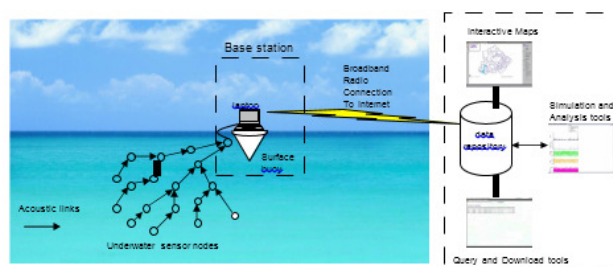


Figure 1. Target network application

The modules periodically sample their sensors, collecting physical indicator data such as temperature and salinity, which influence pollution levels in the water. After sampling their sensors, the nodes report their data to a surface node nearby. The surface node, known as the base station, consists of a portable computer or a personal digital assistant (PDA) with a water-immersed acoustic transceiver

that communicates with the submerged nodes. The computer is also equipped with a long range wireless broadband communication card that uses a cellular or satellite connection. The computer streams the network data towards a central database that acts as the main data repository.

The collected data will feed into a data repository that archives historical data from the monitored area. The proposed network deployment will stream near real time data from the aquatic environment into the data repository, providing professionals in the water management and research communities with access to sizeable and timely data from the water. The application will also leverage existing environmental information management platforms for providing the tools necessary for analyzing, displaying, and sharing the collected data.

The work described in this paper outlines the design considerations for the development of software-driven submerged sensor networks for monitoring pollution indicators in rivers, lakes, estuaries, and coastal areas and subsequently providing the data to environmental engineers in near real-time. While recent surveys cover submerged acoustic networks and practical issues in these networks, the focus here is on software driven submerged sensor networks in shallow water. In particular, this paper leverages our recent field experiments to identify suitable design choices for the acoustic communication link including signal modulation, symbol synchronization, filtering, and demodulation. The paper also visits medium access control (MAC) and routing layer issues and proposes minimalist cross layer design to cope with limited bandwidth and energy resources of submerged sensor networks. Our field experiments have also revealed several logistical issues for submerged sensor network deployments, including waterproofing, casing, calibration, and fouling.

## II. TECHNICAL CONSIDERATIONS

### 2.1 Physical Layer

The physical layer design for our system involves both communication and modulation. The communication hardware is simply the microphone and speaker built-in to sensor nodes, serving as receiver and transmitter respectively. The focus then, is on signal modulation, which takes place through software resident on the motes. Figure 2 shows the block diagram for the communication system. The following discussion explores the main components of Figure 2 separately.

#### 2.1.1 Modulation

The first component of the acoustic communication system is software modulation that takes digital data as input and modulates an acoustic signal with the data. The potential choices of modulation schemes for software modems include amplitude shift keying (ASK), phase shift keying (PSK), and frequency shift keying (FSK). We have selected FSK as the lowest complexity mechanism to run on the resource-limited mote platforms.

In general, FSK uses  $2N$  frequencies to encode  $N$  bits per frequency. The signal demodulation at the receiver can use low complexity techniques, such as the Fast Fourier Transform (FFT), to determine the frequency content of the signal. Choosing a number of frequencies that have high signal-to-noise ratio (SNR) for the channel and ensuring sufficient spectral separation between the frequencies for FSK provides robust submerged communication for low power transmissions with minimal processing complexity at the receiver.

We recently performed an empirical study to investigate the spectral properties of the submerged channel in a controlled water environment. The study used the T mote Invent from Motive module speaker as the transmitter of acoustic signals and a generic PC microphone as the receiver. The components were waterproofed using off-the-shelf elastic latex membranes that vibrate sufficiently to preserve most the acoustic properties of the speaker and microphone. The study evaluated the signal-to-noise ratio (SNR) of frequency tones between 400 and 6700 Hz at 100 Hz increments. The selection of the 100 Hz band separation between frequencies provides for low complexity frequency detection at the receiver.

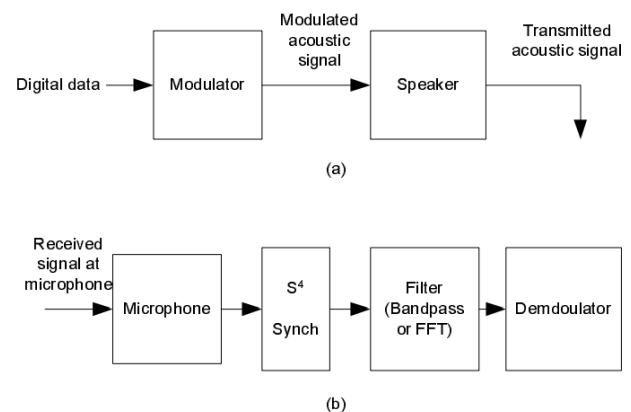


Figure 2. Block diagram for software modem (a) Modulator/Transmitter (b) Demodulator/Receiver

Note that narrower separation bands enable the use of more frequencies within the same available bandwidth, which increases bit rates but also reduces signal quality at the

receiver. In general, there is a tradeoff in digital modulation techniques between the number of signal levels (in phase, amplitude, or frequency) and the quality of the signal. In our case, increasing the number of frequencies by using narrower frequency bands requires the use of higher order filters or fast Fourier transform (FFT) at the mote to decode the signal. This is not feasible with the limited processing power of the motes.

The study revealed that the channel, which includes the speaker, latex membranes, the water, and the microphone, exhibits the highest SNR at frequencies in the range of the 1000 Hz to 2500 Hz. The SNR drops steadily at frequencies above 3 KHz. Building on these results, we designed a software FSK modem that uses 8 frequencies in the 1-2.5 KHz range to represent 8 symbols of 3 bits each. The testing of the modem investigated 4 different bit rates from 6 bps to 48 bps, by varying the symbol duration from 500 ms to 62.5 ms. The experiment results revealed a low bit error rate for all data transfer rates within a transmission distance of 17 m, which was the size of the testing area.

The output of the modulator block is a modulated acoustic signal that is then transmitted over the wireless medium through the built-in speaker.

### 2.1.2 Symbol Synchronization

The transmitted signal is received by the microphone at another node, on which the software then proceeds in the decoding process. Essential to acoustic communication with software modems is the ability of the receiver to synchronize to the first symbol of an incoming data stream. Traditional symbol synchronization approaches rely on the transmission of a predefined sequence of symbols, often referred to as a training sequence. The conventional approach makes two assumptions about the communication channel that do not hold for software-driven acoustic communications, namely: (1) a bit rate at least in the order of tens of kilobits per second; (2) a bit error rate (BER) in the order of  $10^{-6}$  or lower. Software-driven acoustic communication, both aerial and submerged, supports lower bit rates that range between tens to hundreds of bits per second. In addition, the bit error rate of software-driven acoustic communication is several orders of magnitude higher than the radio frequency bit error rate. The higher bit error rate in acoustic communications tends to cause loss of training sequence symbols, preventing proper symbol synchronization. Furthermore, providing high redundancy in the training sequence to mitigate training symbol losses is not an option for the narrow usable bandwidth of software driven acoustic communications.

We recently proposed and tested the S4 synchronization mechanism that uses Short Signature Synchronization

Symbols (S4) to align symbol boundaries at the receiver. The design of the signature symbol aims at a high probability of correlation at the receiver even in cases of partial loss of the symbol and at low probability of false synchronization with ambient noise or data symbols. To this end, the symbol features include two square waves, with each square wave transmitted at a predefined frequency, separated by a predetermined guard time. The use of two frequencies for the signature symbol mitigates the effects of frequency selective fading or interference. The signature symbol guard time duration is chosen so that it is not equal to, and not a constant multiple of, the inter-symbol guard times to avoid high correlation with data symbols. The output of the S4 provides the index of the first data sample in the received signal.

### 2.1.3 Filtering

The receiver then proceeds to filter the received signal. Filtering the acoustic signal at the receiver minimizes the effect of out-of-band noise on the decoding process. A suitable choice of filtering method depends highly on the processing capability of at the receiver. Our system currently provides two filtering methods, with the first method employing narrow band pass filters at the relevant frequencies. The second filtering method applies an FFT to the received signal and examines the signal amplitude at the FFT samples corresponding to the relevant frequencies.

The narrowband filtering method provides a finer signal quality as it only focuses on the frequencies of interest and excludes all interference outside this spectrum. The superior performance of the narrowband filtering method comes at the cost of higher processing activity at the receiver. In fact, the narrowband filtering method is suitable for running on a PC or a PDA that acts as a base station for the submerged network, as in Figure 1.

The FFT method can run on the motes themselves as it has lower processing complexity. Running the FFT method on motes enables the deployment of a multi-hop network where nodes can autonomously decode, process, and relay received signals. The FFT method provides a coarser signal quality than filtering since it does not exclude interference from outside the frequency spectrum of interest.

### 2.1.4 Demodulation

The filtered signal then proceeds to the demodulator component. The demodulator begins examining the signal at the first data sample, which has been determined by the S4 block. Starting at the first data sample and taking the number of samples that corresponds to one symbol, the demodulator determines the frequency component with the highest amplitude within this window, and outputs the data

symbol corresponding to the highest frequency. For subsequent bits, the demodulator shifts the start sample of the previous symbol by the symbol length, and repeats the process of determining the strongest frequency component.

## 2.2 Communication Protocols

The design of the higher layer communication protocols for this system should adopt a minimalist approach to avoid creating high overhead in this bandwidth and energy limited system. The MAC protocol design for submerged sensor networks cannot exploit traditional low duty cycle protocols for terrestrial sensor networks. The reason is that the energy cost of transmission is generally much higher than the cost of signal reception in submerged networks, whereas the cost of transmission and reception is almost the same in terrestrial sensor networks. Thus, signal transmissions dominate the energy consumption profile of submerged nodes. As such, the MAC protocol design should minimize control overhead messages, such as request to send (RTS) and clear to send (CTS), rather than implementing sleep policies. One possibility for our system is to use burst tones at the beginning of transmission to reserve the channel. In particular, the S4 preamble can serve as a burst tone for reserving the channel at the MAC layer.

For routing data towards the surface node, our communication system advocates the use of short multihop links for deploying dense submerged sensor networks. Keeping in line with the system's minimalist approach, the network should make use of simple multihop routing techniques, such as the directed broadcast with overhearing method of MERLIN. The relatively low cost of receiving signals actually favors the use of overhearing. Our recent study on the upper bounds for transmissions in directed broadcast networks with overhearing has shown that this method has only few redundant packet, resulting in low overhead for large networks.

The proposed communication protocol design concepts exploit cross-layer interactions to optimize the use of bandwidth and energy resources for this system. We expect that further cross-layer optimization opportunities will arise as the project advances.

## III. LOGICAL CONSIDERATIONS

### 3.1 Waterproofing and Casing

The most common waterproofing method for submerged communications hardware is to place the hardware in a custom-designed waterproof case with a special air-locked hole for the hydrophone and transducer that are in contact with the water. The cost associated with the material and design of the custom-designed casing strategies

significantly increases system cost and discourages large scale deployments of submerged sensor networks.

Our project's design strategy advocates the use off the-shelf components not just for communication and sensing, but also for the protection and waterproofing of the components. As such, our design places the sensor nodes in elastic latex membranes that are cheap and readily available on the market. The latex membranes take the form of a balloon that is sealed to waterproof the sensor nodes. Since the sensor module includes the speaker and microphone, these acoustic communications components are also fully enclosed within the latex membranes. The elasticity of the membranes ensures that the acoustic waves transmitted by the speaker are transferred to the water through the elastic membrane. At the receiver side, the membranes vibrate upon the reception of an acoustic signal, transferring the signal to the interior of the membrane where the microphone can detect it.

Transfer rate (bps)	6	12	24	48	96
Latex mem. Up to 17m	≥95%	≥90%	≥81%	≥79%	N/A
Vinyl mem. Up to 10m	N/A	≥90%	≥78%	≥35%	≥10%

Table 1. Comparison of the percentage of symbols correctly received for the PC speakers and Tmote Invent experiments

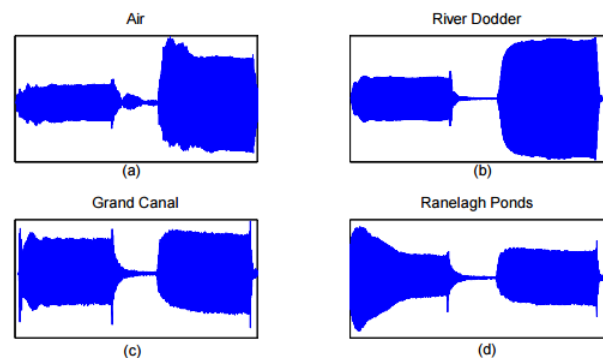


Figure 3. Shape of S4 symbol as received in (a) Indoor aerial channel; (b) River Dodder; (c) Grand Canal in Dublin; (d) Ranelagh pond.

The use of the latex membranes causes relatively small reductions in signal amplitude. Our recent study compared the suitability of two membranes for waterproofing the sensor nodes: (1) a vinyl membrane; and (2) a latex membrane. Table 1 illustrates the percentage of correctly received symbols for bit rates ranging between 6 bps and 96 bps for both membranes. The vinyl membrane experiments

were conducted with a generic PC speaker as a transmitter, while the latex membrane experiments were conducted with Tmote Invent speakers. The output power rating of the two speaker types is the same, providing solid ground for comparing the results of the two experiment sets. The results in Table 1 show that the latex membrane has a better coupling with the water, yielding a notably lower bit error rate at the higher transmission rate of 48 bps.

### 3.2 Calibration

Our field tests in different bodies of water have revealed a distinct background noise and interference pattern in each case. For instance, the primary noise source in swimming pools is water pumps, whereas the noise sources in a river include currents and wildlife activity. The dependence of the noise profile on the deployment environment requires calibration steps, which could be manual or automatic, prior to placing the sensors in the water. Fortunately, the implementation of modulation and communication in software provides maximum flexibility for on-the-fly calibration.

A central issue for calibration is the frequency selective noise in the deployment environment. The choice of frequencies for the S4 synchronization symbol must avoid frequencies with high noise in a particular deployment environment. The selection of the S4 frequencies is critical for proper system operation, as choosing unsuitable frequencies causes large synchronization errors, resulting in many bit misalignments. Similarly, it is also important to choose data symbol frequencies that avoid the high noise frequencies.

The selection of the suitable frequencies for S4 synchronization and the data symbols can be done automatically. During an initial setup phase, one node, typically the base station, can be designated as a calibration receiver, and another as a calibration transmitter. Upon deployment, the designated transmitter sends an a priori known calibration signal that includes a diverse set of S4 symbols with different frequency combinations, followed by a sequence of frequency tones that covers all the possible symbol frequency tones. The designated receiver processes the calibration signal by comparing the processed signal against a locally stored version of the reference signal. The receiver then selects the frequencies that have been received with the highest SNR, and transmits a short broadcast message indicating these frequencies to the other nodes.

Another calibration issue is ensuring that the stored S4 symbol at each node is representative of the symbol as it is received in the current deployment environment. Our field experiments have shown that the structure of a received S4 symbol with the Tmote Invent microphone in air is different

than the structure in water. As Figure 3 shows, the structure even varies across different water media, depending on the current, depth, and suspended solids in the water. For instance, the envelope of the S4 symbol differs significantly when the signal is received in the river (Figure 3(b)) and in the pond (Figure 3(d)). The plots in Figure 3 illustrate that the relative amplitude of each of the square signals changes depending on the medium. Other signal artifacts, such as impulses at the beginning or end of the signal, are also dependent on the deployment environment. As such, each node should store at least one instance of the S4 symbol as it is received in the current deployment environment. This maximizes the probability of successful correlation and synchronization through the S4 mechanism.

### 3.3 Fouling

Fouling is a process by which marine wildlife, such as barnacles, zebra mussels, weeds, and algae attach themselves to still object in the water. For this project, avoiding fouling effects is especially important since the attachment of organisms to the membrane could limit or change the membrane's vibration characteristics. Traditional anti-fouling approaches include the use of special copper-based paints to prevent the attachment of organisms to boat bottoms. Current research focuses on developing alternatives to paint based solutions, which are harmful to the ambient environment.

Our project aims at protecting the environment, and not damaging it in the process, so we intend to adopt one of the emerging anti-fouling techniques. One interim solution under consideration is to place the nodes in the latex membranes and then to fix the membranes inside a resilient cubic plastic box whose purpose is to shield the nodes from fouling and other hazards in harsh submerged environments. The surfaces of the plastic box would be perforated to maintain acoustic coupling between the box contents and the water.

## IV. CONCLUSION

This paper has presented the lessons learned so far from our field experiments with software-driven submerged sensor networks. The main technical challenges include the design and integration of the modulation, synchronization, filtering, and demodulation techniques for the software modems. The design of higher layer communication protocols for this system should also adopt a minimalist approach to minimize control overhead. Logistical considerations for submerged networks depend on the deployment environment. We have identified common logistical considerations low-power submerged networks,

which include the need for resilient waterproofing and casing, calibration, and anti-fouling measures. Our initial experiments in various bodies of water have exposed the benefits of using software modems for submerged communications capable of functioning on off-the-shelf multipurpose sensors. The design guidelines in this paper lay the groundwork for further development of software driven of submerged sensor networks.

## References

- [1] Yu Yang; Coll. Of Marine, Northwestern Polytech. Univ., Xi'an; Zhang Xiaomin ; Peng Bo ; Fu Yujing, "Design of sensor nodes in submerged sensor networks", Published in: Industrial Electronics and Applications, 2009. ICIEA 2009. 4th IEEE Conference on Date of Conference: 25-27 May 2009 Page(s): 3978 – 3982.
- [2] Dharan, K.T. ; SITE, VIT Univ., Vellore, India; Srimathi, C. ; Soo-Hyun Park, "A sweeper scheme for localization and mobility prediction in submerged acoustic sensor networks", Published in: OCEANS 2010 IEEE – Sydney Date of Conference: 24-27 May 2010 Page(s): 1 – 7.
- [3] Peng Jiang ; Syst. Eng. Res. Inst., CSSC, Beijing, China, "An submerged sensor network localization algorithm based on Time-of-Arrivals (TOAs)", Published in: Natural Computation (ICNC), 2013 Ninth International Conference on Date of Conference: 23-25 July 2013 Page(s): 1552 – 1556.
- [4] Donghoon Kim; Dept. of Comput. Sci., Florida State Univ., Tallahassee, FL ; Yong-Man Cho ; Changhwa Kim ; Sangkyung Kim, "E-ITRC Protocol with Long & Adjustable Range on Submerged Acoustic Sensor Network", Published in: Advanced Information Networking and Applications Workshops, 2007, AINAW '07. 21st International Conference on (Volume:2) Date of Conference: 21-23 May 2007 Page(s): 665 – 672.
- [5] Yuh-Shyan Chen; Dept. of Comput. Sci. & Inf. Engineering, Nat. Taipei Univ., Taipei, Taiwan ; Yun-Wei Lin; Sing-Ling Lee, "A mobicast routing protocol in submerged sensor networks", Published in: Wireless Communications and Networking Conference (WCNC), 2011 IEEE Date of Conference: 28-31 March 2011 Page(s): 510 – 515.
- [6] Blyskun, A. ; TrellisWare Technol., Inc., San Diego, CA, USA ; Johnson, M. ; Sungill Kim ; Speros, J., "Improving the SRW Waveform via a Physical Layer Retrofit", Published in: Military Communications Conference, MILCOM 2013 - 2013 IEEE Date of Conference: 18-20 Nov. 2013 Page(s): 396 – 399.
- [7] Vinodhini, C. ; M.E Commun. Syst., Anand Inst. of Higher Technol., Chennai, India ; Sivasankar, S., "Back to Results Improving wireless physical layer security via co-operative techniques" Published in: Innovations in Information, Embedded and Communication Systems (ICIIECS), 2015 International Conference on Date of Conference: 19-20 March 2015 Page(s): 1 – 4.
- [8] Sengezer, N. ; Dept. of Electr. & Electron. Eng., Bilkent Univ., Ankara, Turkey ; Karasan, E., "Multi-layer Virtual Topology Design in Optical Networks Under Physical Layer Impairments and Multi-hour Traffic Demand", Published in: Optical Communications and Networking, IEEE/OSA Journal of (Volume:4 , Issue: 2 ) Page(s): 78 – 91.
- [9] Lau, V.K.N. ; Dept. of Electr. & Eng., Univ. of Hong Kong, China ; Yu-Kwong Kwok, "On the synergy between adaptive physical layer and multiple-access control for integrated voice and data services in a cellular wireless network", Published in: Vehicular Technology, IEEE Transactions on (Volume:51 , Issue: 6 ) Page(s): 1338 – 1351 ISSN : 0018-9545.
- [10] Qingyang Song ; Northeastern Univ., Shenyang, China ; Lei Guo ; Fanzhao Wang ; Shiqiang Wan, "MAC-centric cross-layer collaboration: a case study on physical-layer network coding", Published in: Wireless Communications, IEEE (Volume:21 , Issue: 6 ) Page(s): 160 – 166 ISSN : 1536-1284 .
- [11] Ivanchenko, M.V. ; Dept. of Radiophys., Nizhny Novgorod Univ., Russia ; Osipov, G.V. ; Shalfeev, V.D., "Synchronization of chaotic oscillators with type-I intermittency", Published in: Physics and Control, 2003. Proceedings. 2003 International Conference (Volume:2 ) Date of Conference: 20-22 Aug. 2003 Page(s): 563 - 568 vol.2.
- [12] Xin Qian ; Inst. of Comput. Applic. Technol., Hangzhou Dianzi Univ., Hangzhou, China ; Xinfu Shen ; Guojun Dai ; Jianhui Zhang, "Clapping and Broadcasting Synchronization in Wireless Sensor Network", Published in: Mobile Ad-hoc and Sensor Networks (MSN), 2010 Sixth International Conference on Date of Conference: 20-22 Dec. 2010 Page(s): 140 – 145.
- [13] PalChaudhuri, S.; Dept. of Comput. Sci., Rice Univ., Houston, TX, USA; Saha, A.K.; Johnsin, D.B., "Adaptive clock synchronization in sensor networks", Published in: Information Processing in Sensor Networks, 2004. IPSN 2004. Third International Symposium on Date of Conference: 26-27 April 2004 Page(s): 340 – 348.
- [14] Zhenping Chen ; Sch. of Electr. & Inf. Eng., Anhui Univ. of Sci. & Technol., Huainan, China ; Yourui Huang ; Dequan Li, "One feedback-based time synchronization method for wireless sensor networks", Published in: Control and Decision Conference (CCDC), 2015 27th Chinese Date of Conference:23-25 May 2015 Page(s): 1459 – 1464.

- [15] Xiujuan Dou ; Sci. & Technol. on Electron. Inf. Control Lab., Chengdu, China ; Yujun Kuang ; Liang Jiao ; Feng Yang, "A novel parallel state search based synchronization scheme for frequency hopping system", Published in: Computational Problem-Solving (ICCP), 2012 International Conference on Date of Conference: 19-21 Oct. 2012 Page(s): 418-422.