



Experimental Study of DQPSK Modulation on SDR Platform

^{1,2}Eko Marpanaji, ²Bambang Riyanto T., ²Armein Z.R. Langi, ²Adit Kurniawan,
²Andri Mahendra, ²Thay Liung

¹Departement of Electronics Engineering, Yogyakarta State University
Karangmalang, Yogyakarta 55281, Telp. +62-274-586168, Indonesia

²School of Electrical Engineering and Informatics, Institut Teknologi Bandung
Jl. Ganesa 10 Bandung 40132, Telp. +62-22-2501661, Fax. +62-22-2534133, Indonesia

Abstract. This Paper addresses Differential Quadriphase Shift Keying (DQPSK) modulation implemented on SDR platform for the development of digital data communications based on SDR. DQPSK modulation performance perceived at Packet Error Rate (PER) is evaluated in terms of E_b/N_0 or S/N ratio, carrier frequency, bit rate, gain, roll-off factor of root Nyquist filter or root raised cosine filter, and of size of payload from delivered data. Based on this results, the smallest PER could be obtained by setting E_b/N_0 value which is greater than 20 dB, carrier frequency of at least 0,3 MHz, optimum bit rate of 200 kbps, optimum range payload size of 2000 up to 4000 bytes, and roll-off factor of Nyquist or root-raised cosine filter of 0.4.

Keywords: *Differential quadriphase shift keying, packet error rate, root raised cosine, software-defined radio, usrp.*

1 Introduction

This paper describes the implementation of *Differential Quadriphase Shift Keying* (DQPSK) modulation scheme for communicating of digital data on *Software-Defined Radio* (SDR) platform. This performance evaluation of DQPSK modulation scheme is also as a part of the current research in building SDR platform which can be reconfigured, including answering the research question of “How to design SDR platform using commodity components.”

Chen (2007) investigated a performance of *Quadriphase Shift Keying* (QPSK) modulation scheme to send a video digital data. All algorithms are modeled and simulated in Matlab, and the future works of that research is integrated those model into GNU Radio SDR platform using USRP[1]. Thus, the investigation of QPSK modulation scheme performance and its variants, including DQPSK, on SDR platform using *Universal Software Radio Peripheral* (USRP) is still needed.

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In this research, we investigated the performance of DQPSK modulation scheme implemented on SDR platform using USRP from GNU Radio. DQPSK modulation scheme is chosen in this research because the QPSK and DQPSK modulation is also widely used in recent wireless and cellular communication systems such as Satellite, CDMA, and cable modem.

The rest of this paper is organized as follows. Section 2 discusses about SDR architecture along with SDR's characteristics. Section 3 presents the DQPSK modulation scheme. Section 4 presents system configuration used in this research. Section 5 presents DQPSK implementation and test result. Finally, Section 6 is conclusion.

2 Software-Defined Radio (SDR)

Software-defined radio (SDR), sometimes shortened to software radio (SR), was introduced for the first time in 1991 by Joseph Mitola[2]. The word of SDR was used to show a radio class that could be re-configured or re-programmed[3], thus resulted a kind application of wireless communication with mode and frequency band determined by software function. Ideally, SDR offers flexibility, re-configurability, scalability and as multi mode as possible.

SDR architecture is developed based on conventional radio functions. The difference is all functions of signal processing on conventional radio are carried out fully by hardware while the functions of signal processing on SDR are carried out as much as possible by software. The major key in building SDR is the placement of ADC and DAC components as a divider between analog and digital domain, thus the signal processing can be carried out using software.

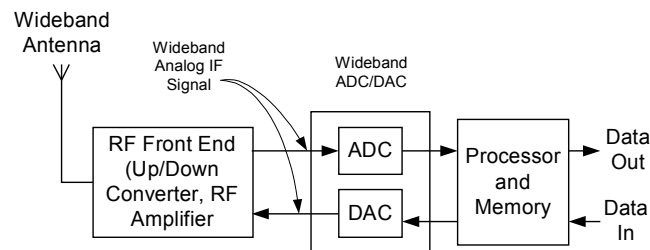


Figure 1Realistic SDR Architecture

The ideal SDR architecture will place ADC/DAC as close as possible with the antenna, and will be followed by processor to carry out digital signal processing. The radio functions are carried out by the software which is operated by the current processor. Therefore, if there is standard change of

communication, thus its change can be anticipated by upgrading the software[2][3]. However, the technology limitation and the expensive ADC/DAC wideband urge to change a little of SDR architecture in placing ADC/DAC. The more realistic SDR architecture places ADC/DAC wideband after Down Converter/Up Converter as shown in Fig. 1. With this architecture, the conversion from analog to digital or reverse is carried out on Intermediate Frequency (IF) signal which possesses lower frequency than RF signal. Today, that type of architecture are being developed widely and researched for the implementation. The SDR architecture for both transmitter and receiver can be presented with block diagram as shown in Fig. 2.

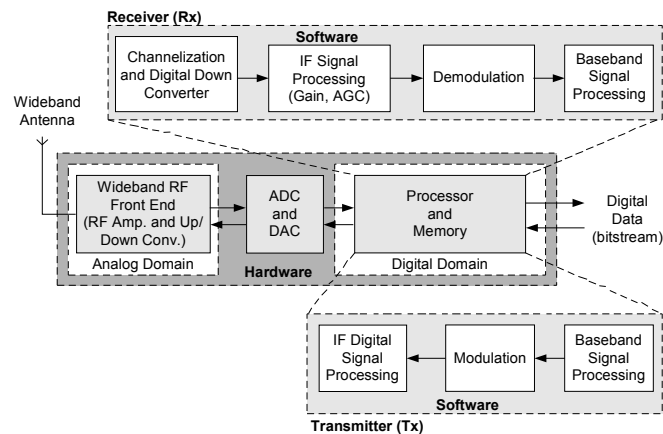


Figure 2 SDR Architecture for Transmitter and Receiver

The SDR platform performs transmitting and receiving functions. The transmitter (Tx) will perform some process such as base band signal processing, modulation, digital IF signal processing, and sending RF signal to the air. The receiver (Rx) will perform some process such as RF signal processing, channelization, digital IF signal processing, demodulation, and base band signal processing. As shown in Fig. 2, the computation process in the receiver will be more complex than the transmitter.

ADC and DAC component will determine a point in which radio functions can be performed through a software. This point is usually called digital access point. The realistic SDR architecture will use software to process the digital IF signal. The major challenge in choosing ADC's location is the problem of limited sampling frequency and economics factor. If the sampling frequency of Nyquist is hard to be done because of its digitalization IF signal is heavy, then the alternative is to apply under sampling toward IF signal, therefore it reduces

the computation complexity from processor. The formula to determine under sampling frequency is as shown below:

$$\frac{2f_c + B}{m + 1} \leq f_s \leq \frac{2f_c - B}{m}; m \in N \quad (1)$$

Where f_c is center frequency, B is bandwidth, f_s is sampling frequency chosen until $f_s > 2B$ [4]. For example, the SDR system uses 20 MHz IF signal with 5 MHz bandwidth can be sampled with frequency of 22,5 MHz for $m=1$, or 17,5 MHz for $m=2$, or 11,66 MHz for $m=3$.

The next attention is on computation complexity which will determine the type and size of processor needed. The processor will generate the software of digital signal processing both for the receiver and transmitter. Besides to observe the functions generated by the software in determining computation complexity, the scalability factor for next development anticipation is also observed. All those steps will determine the processor's size usually stated with *Million Operations per Second* (MOPS) or *Million Instructions per Second* (MIPS).

The biggest computation requirement is on the receiver side especially on filter part in channelization, which range from 100 to 200 operation/sample[3][5][6]. By using the sampling speed of 22.5 MHz will result 22.5 MSPS, thus it needs 2,250 until 4,500 MIPS. Other computation requirements are modulation and demodulation process. As a description, modulation and demodulation FM require more or less 100 to 150 MIPS[7]. The total of these computation requirements will determine the type and the number of processor used in SDR platform. It means, for bigger computation requirement and unable to be carried out by a processor, thus the SDR architecture can use parallel computation architecture by involving several processors or implementing a part of signal processing function in form of FPGA.

In the next section, the SDR implementation for digital data communication using Differential Quadrature Phase Shift Keying (DQPSK) modulation scheme along with the test result to observe its PER will be explained.

3 Differential Quadrature Phase Shift Keying (DQPSK) Modulation

DQPSK modulation is a variant of $\pi/4$ -QPSK modulation scheme, which differential encoding is added to the bits encoding prior to the modulation. Differential means that the information is not carried by the absolute state; it is carried by the transition between states. The advantage of using differential encoding is free from phase ambiguity if the constellation is rotated by some

effect in the communications channel which the signal passes through. This problem can be overcome by using the data to change rather than set the phase. The QPSK modulation scheme using differential encoding is usually called as $\pi/4$ -DQPSK or DQPSK.

The equation differential encoder to produce symbols is:

$$s_k = s_{k-1} \oplus b_k \quad (2)$$

where s_k is current output symbol, s_{k-1} is previous output symbol, and b_k is current bit input. The \oplus sign indicates modulo-2 operation. So s_k only changes state (from binary '0' to binary '1' or from binary '1' to binary '0') if b_k is a binary '1'. Otherwise it remains in its previous state. At the receiver, the received signal is demodulated to yield $s_k = \pm 1$ and then differential decoder reverses the encoding procedure (decoding process) and produces:

$$b' = s_k \oplus s_{k-1} \quad (3)$$

since binary subtraction is the same as binary addition. Therefore, $b'_k = 1$ if s_k and s_{k-1} differ and $b'_k = 0$ if they are the same. Hence, if both s_k and s_{k-1} are *inverted*, b_k will still be decoded correctly as b'_k . Thus, the 180° phase ambiguity does not matter.

The DQPSK modulation format uses two QPSK constellations offset by 45 degrees ($\pi/4$ radians) as depicted in Fig. 3. Transitions must occur from one constellation to the other, or each DQPSK symbol will reside in one of eight points in the constellation diagram. This guarantees that there is always a change in phase at each symbol, making clock recovery easier. The data is encoded in the magnitude and direction of the phase shift, not in the absolute position on the constellation.

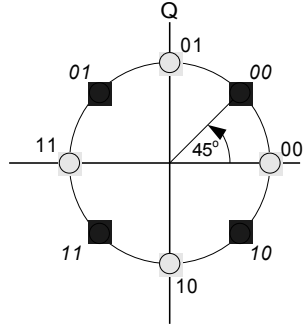


Figure 3 DQPSK constellation diagram with identical Gray coding rotated by 45°

Table 1 summarizes a possible set of relationships between the phase transitions in the DQPSK modulation scheme and the incoming Gray code dibits[8].

Table 1 Correspondence between input dibit and phase change for DQPSK modulation

<i>Gray-Encoded Input Dibit</i>	<i>Phase change, $\Delta \theta$ (radians)</i>
00	$\pi / 4$
01	$3\pi / 4$
11	$- 3\pi / 4$
10	$- \pi / 4$

DQPSK used root Nyquist filter or root raised cosine filter as a pulse-shaping filter. This filter is used to reduce a number of spurious signals or to control the shape of digital data. The impulse response of root Nyquist filter is determined by equation [3]:

$$g(t) = \frac{1}{\pi t} \frac{1}{1 - \left(\frac{4\alpha t}{T_b}\right)^2} \sin\left\{2\pi(1 - \alpha)\frac{1}{T_b}\right\} + \frac{1}{\pi} \frac{4\alpha / T_b}{1 - \left(\frac{4\alpha t}{T_b}\right)^2} \cos\left\{2\pi(1 + \alpha)\frac{1}{T_b}\right\} \quad (4)$$

where α is roll-off factor of root raised cosine filter. This parameter controls the shape and bandwidth of the incoming signal.

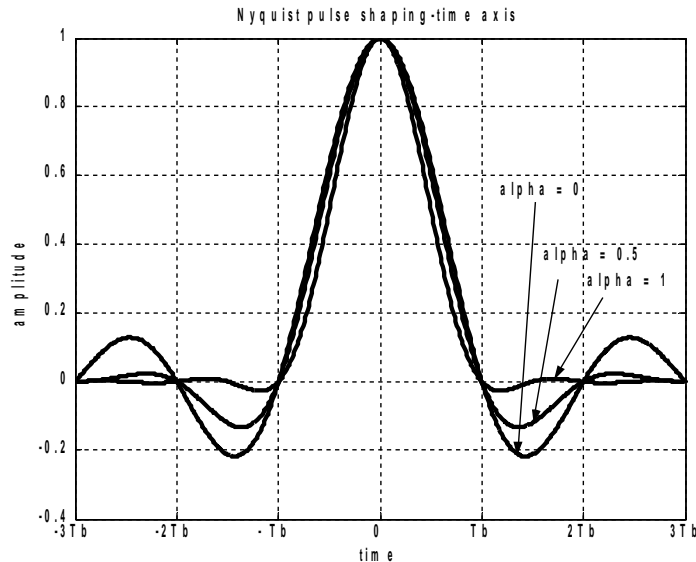


Figure 4 Impulse response of a Nyquist filter in time domain with several chosen roll-off factors.

Fig. 4 shows the impulse response of a root Nyquist filter in time domain with several chosen of roll-off factor (α) values. One features of the Nyquist filter is that we always obtain impulse response is zero at nT_b (n is integer: 1, 2, 3, ...) in the time domain. Therefore, when we set the synchronization point at nT_b , one symbol never interferes with other symbols at this point.

The advantages of DQPSK are:

- the phase transitions from one symbol to the next are restricted to $\pm \pi/4$ and $\pm 3\pi/4$ so it will reduce the symbol ambiguity
- using root raised cosine filtering, it has better spectral efficiency than *Gaussian Minimum Shift Keying* (GMSK), the other common cellular modulation type.

3.1 DQPSK Modulator

Fig. 5 shows the principle of a DQPSK modulator. The process occurred on modulator starts by changing bit stream with level 0 and 1 into bipolar stream bit with level -1 and 1. Next, bipolar stream bit is processed by mapping circuit to produce dibits for *I-channel* (Ich) and *Q-channel* (Qch), and then each dibit data will be processed by differential encoder using the rules of Table 1 separately. The output of differential encoder will be passed through a pulse-shaping filter to reduce its spurious signals. Finally, the output of pulse-shaping filter is converted to analog signal and modulated using RF signal $\cos(2\pi f_c t)$ for Ich signal and $\sin(2\pi f_c t)$ for Qch signal, where f_c is carrier frequency. Both Ich and Qch signal are combine together to produce DQPSK signal.

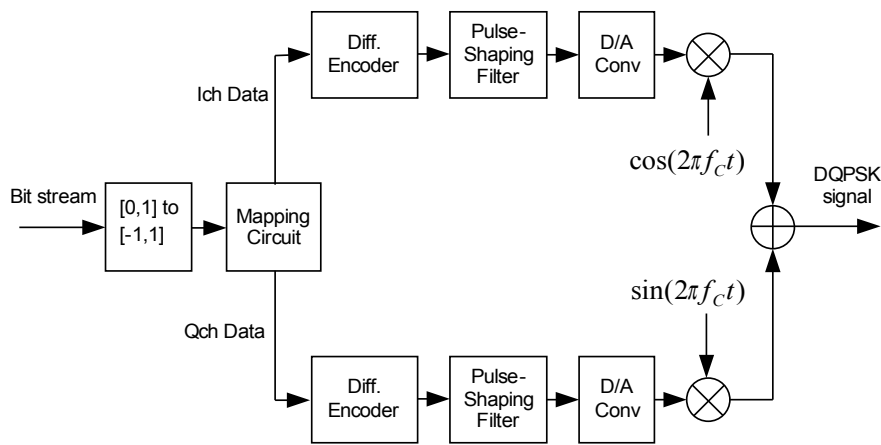


Figure 5 DQPSK Modulator

3.2 DQPSK Demodulator

At receiver or demodulator of DQPSK, process that occurred is reverse at modulator of DQPSK. Fig 6 showing block diagram of DQPSK demodulator.

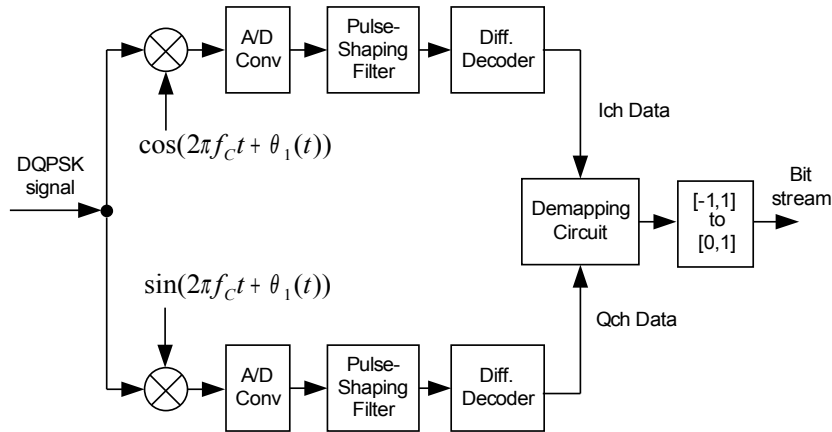


Figure 6 DQPSK Demodulator

Referred to the figure, originally demodulator of DQPSK will eliminate carrier frequency so that recover Ich and Qch signal. Both signal then converted in the form of digital signal through A/D Converter. Digital signal which yielded then passes through a pulse-shaping filter, and by using differential decoder dibit information transmitted by modulator will be recovered. Dibit information will be dissociated and become consecutive bit through demapping circuit, and bit stream with level - 1 and 1 will be returned to bit stream with level 0 and 1. Thus, consecutive information bit data transmitted by modulator of DQPSK can be found again at DQPSK demodulator.

4 System Configuration

System configuration which used in this research is shown at Fig 7. SDR platform in this research use USRP GNU Radio peripheral that function is up/down converter as a front end and personal computer (PC) as the processor. System configuration of SDR using PC as the processor was often referred as Software Radio or shortened by SWR. Pursuant to configuration, hence modulator and demodulator of DQPSK can be implemented in software, while A /D and D/A converter remain in hardware form which located in USRP. USRP and PC is connected using port USB version 2.0.

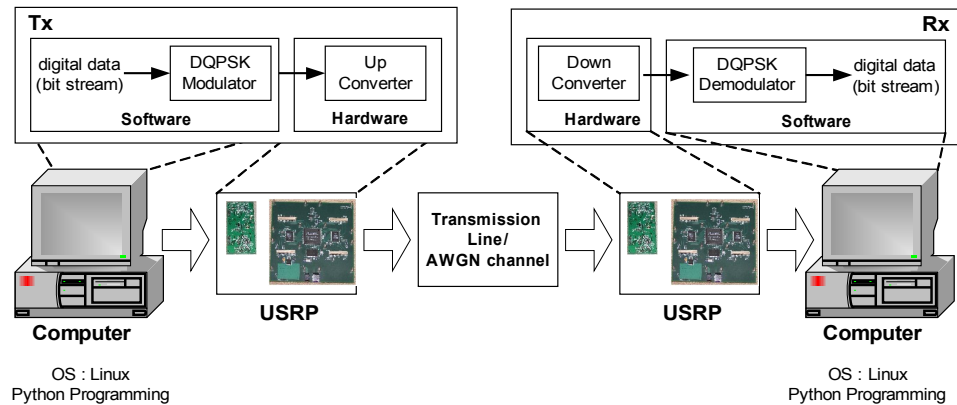


Figure 7System configuration of SDR platform

Mainboard specification of USRP which used in this research is: (a) USB 2.0 port for connection to computer; (b) 12-bit ADC with speed of sampling 64 MSPS so that with principle of aliasing can conduct digitalization process with range frequency of aliasing - 32 MHz until 32 MHz; (c) 14-bit DAC with clock frequency 128 MSPS so have Nyquist frequency equal to 64 MHz; and (d) analog signal which yielded limited to 10 mWatt. While daughterboard using Basic Tx and Basic Rx so that there is no process of up/down converter and transmitter frequency limited 50 MHz maximum.

Specification of PC which used in this research is summarized at Table 2. Computer that used must have USB version 2.0 port to support connection with USRP board. Programming language used Python and a few block functions written in C++.

Table 2Computer specifications that used in this research

No	Component	1 st Computer	2 nd Computer
1.	Processor	AMD Athlon XP1800+, 1.53 GHz, fsb 533 MHz	Intel Pentium 4, 2.93 Ghz, fsb 533 MHz
2.	RAM	DDR 333 MHz, 256 Mbyte	DDR 400 MHz, 2 x 256 MByte
3.	Operating System	Linux Fedora Core-4, 2.6.11-1.i369	Linux Fedora Core-4, 2.6.11-1.i369

System performance can be measured using one of the parameter which was often used in QoS, that is:

$BER < 10^{-3}$, PLR or $PER < 10^{-2}$, spread delay < 100 ms, and $Gos > 95\%$ [4].

In this research performance modulation scheme of DQPSK is measured in Packet Error Rate (PER) because data transmitted in the form of packet. Transmitting data as a packet will facilitate of SDR software integration with TCP/IP protocol for next development.

5 Implementation and Test Result

DQPSK modulator and demodulator described previously are implemented in SDR platform using PC and USRP board that commonly used in SDR experiment from GNURadio. The Board is a front side of SDR system and has functions to execute digitalization function (ADC-DAC) and filter channelization process. The digital signal processing used personal computer with USB port as a link between USRP and PC. Operating system used Linux with python programming language to develop the application. All software used for digital signal processing was open source from GNU Radio.

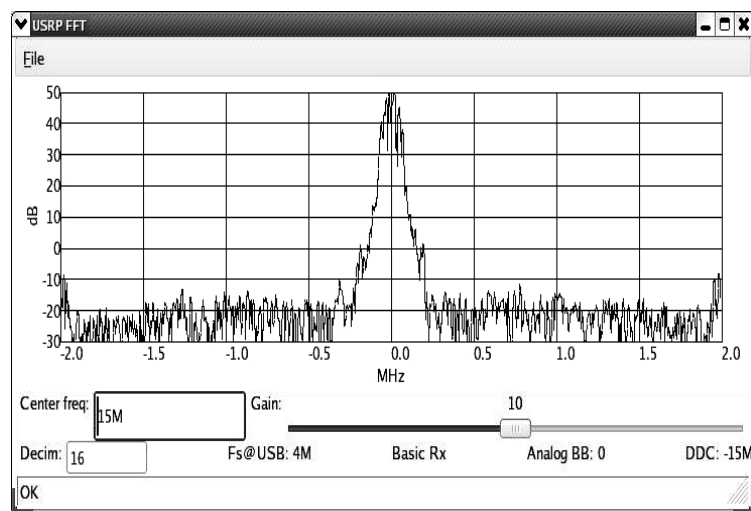


Figure 8 Spectrum DQPSK signal

This experiment would observe the performance of digital communication based on SDR system using USRP board with DQPSK modulation scheme. The transmission channel of coaxial cable RG58 was used to connect transmitter and

receiver because of limitation of Basic TX output power. The experiment was carried out to observe Packet Error Rate (PER) stated by the percentage of exactly accepted packet toward variation of Eb/No, bit rate, carrier frequency, roll-off factor, gain, and also payload size.

Fig. 8 shows the spectrum of DQPSK signal which has 15 MHz carrier frequency, bit rate 256 kbps, and roll-off factor 0.35. The result of observation is shown in Fig. 9 up to Fig. 14.

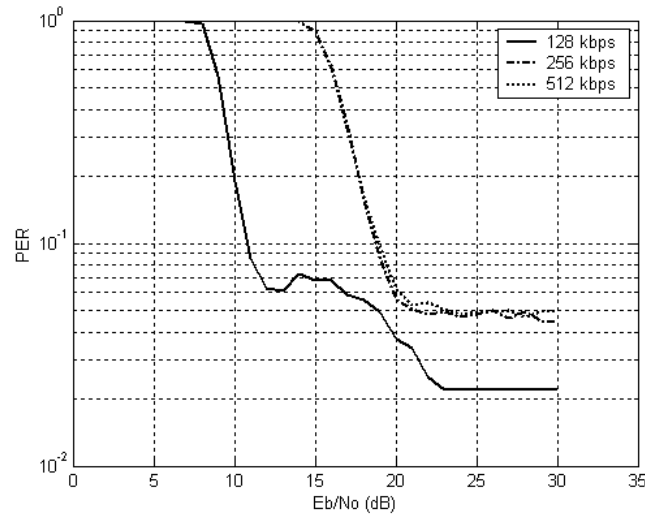


Figure 9 The effect of Eb/No on PER

Fig. 9 shows the graph between variation of Eb/No concerning the PER value with 128 kbps, 256 kbps, and 512 kbps bitrate. Referred to the graph perceived, greater value of Eb/No hence smaller PER will be obtained. PER value will remain low for the value of Eb/No higher than 20 dB.

Variation of frequency carrier modulations of DQPSK will affect to PER value as depicted in Fig 10. The Graph obtained by arranged bit-rate transmitted data equal to 128 kbps, 256 kbps, and 500 kbps. In this experiment maximum frequency that tried is 44 MHz, and beyond that frequency cannot be handled by USRP using existing Basic TX. Referred to the graph, lowest PER value can be obtained if $f_c > 0.06$ MHz for 128 kbps, $f_c > 0.2$ MHz for 256 kbps, and $f_c > 0.3$ MHz for 500 kbps. In another word, the minimum frequency which support for transmitting data with bit-rate greater than 256 kbps is 0.3 MHz.

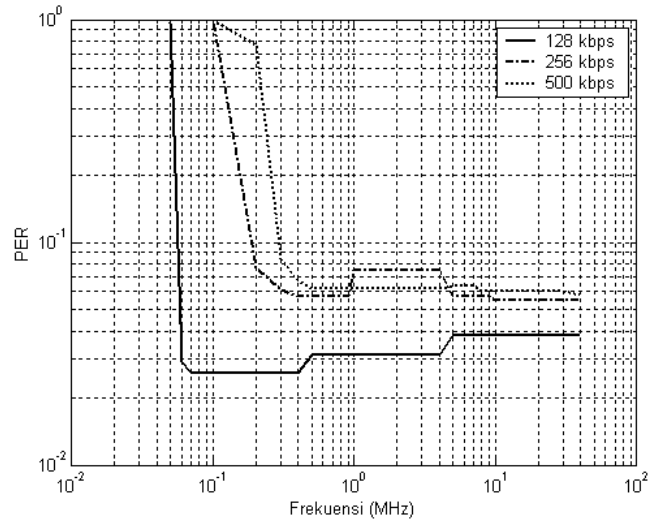


Figure 10 The effect of carrier frequency on PER

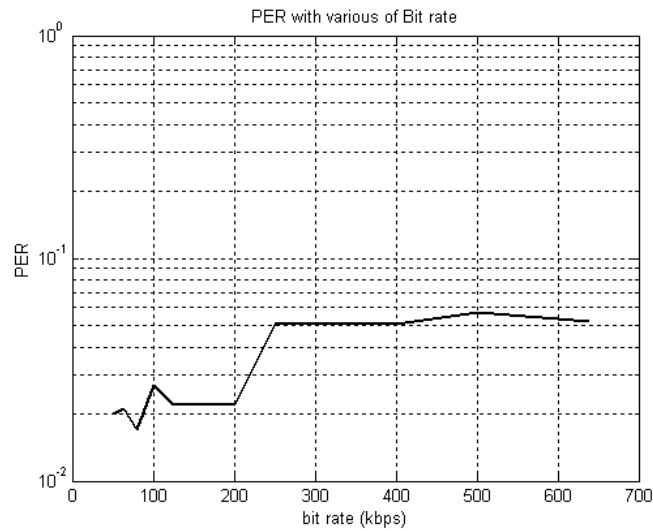


Figure 11 The effect of bit rate on PER

Fig. 11 shows a graph of bit-rate variation to PER value. Referred to the graph, the optimum bit-rate for the modulation of this DQPSK is 200 kbps. Bit rate

greater than 200 kbps will increase the PER value and bit rate from 250 up to 600 kbps produced higher PER value.

Fig. 12 shows a graph of roll-off factor variation value at Nyquist filter to PER value. Referred to the graph, in general can be expressed that bigger roll-off factor value will result smaller PER, and optimum value for roll-off factor to get PER is 0.4. Roll-off factor value for the Nyquist of filter is $0 \leq \alpha \leq 1$, which value of roll-off equal to 1 referred as roll-off full-cosine which is very useful in signal timing extraction for the process of synchronization. Ever greater value of roll-off factor hence smaller influence of ISI.

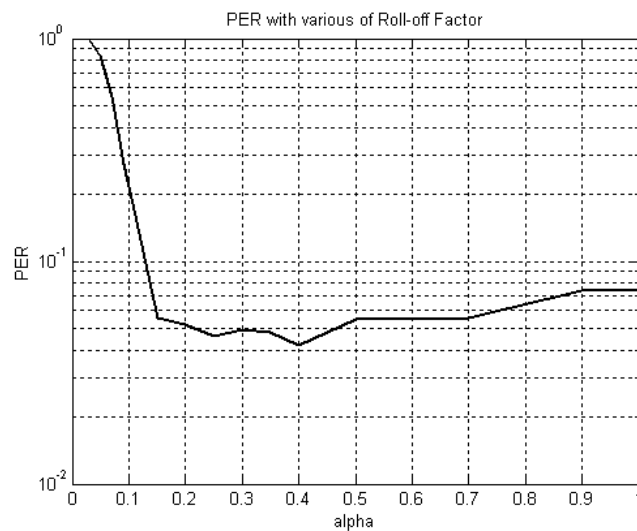


Figure 12 The effect of roll-off factor on PER

Fig. 13 shows a graph between the variation of gain value and resulted PER value. Referred to the graph, the greater gain will result the smaller PER value. However, for the gain value bigger than 500 PER value do not change the PER value. Hence, the optimum value for gain is 500.

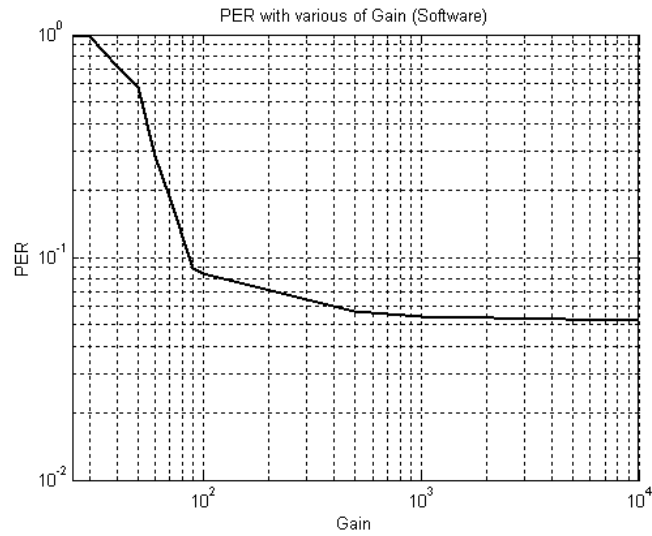


Figure 13The effect of gain PER

Fig 14 shows a graph concerning with variation of payload value to PER value. The graph indicated that the bigger transmitted payload value will produce the smaller PER. The maximum data size for this system is 4092 bytes, hence the optimum value for the measure of payload is 4092 bytes.

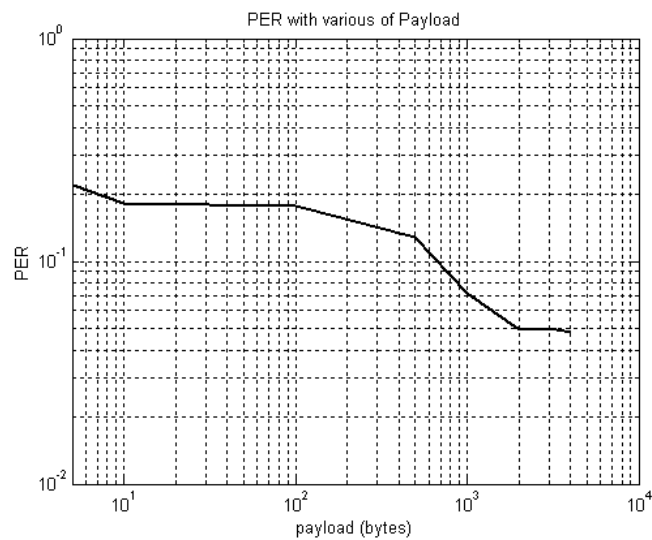


Figure 14The effect of payload size on PER

Referred to all the graphs of this research result, PER value which obtained with variation of Eb/No, carrier frequency, bit rate, roll-off factor, gain, and payload is very low that is closing 10^{-2} . In spite of comparison with criteria measurement performance where $PER < 10^{-2}$ not yet fulfilled. Hence, modulation scheme implementation of DQPSK at platform of SDR use USRP of GNU Radio still require to be completed either software and hardware so that we can get better performance of DQPSK modulation scheme on SDR platform using USRP GNU Radio.

6 Conclusion

Referred to this research result we can conclude that: In general can be said that modulation scheme of DQPSK which implemented at SDR platform using USRP from GNU Radio have better performance which is seen from PER value. Lowest PER Value can be obtained at Eb/No value > 20 dB, optimum bit-rate 200 kbps, carrier frequency bigger than 0,3 MHz, roll-off factor (alpha) is 0.4, optimum gain is 500, and optimum payload value is 4092 byte.

Continuation of this research is software development of SDR to improve the performance and also create application that adept to integration with communications protocol for example TCP/IP.

7 Acknowledgment

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8 References

- [1] Z. Chen., "Performance analysis of channel estimation and adaptive equalization in slow fading channel," 2007, [On-line], Available at http://www.wu.ece.ufl.edu/projects/channelEstimation/full_files.zip.
- [2] J. H. Reed, Software Radio: A Modern Approach to Radio Engineering. New Jersey: Prentice Hall, 2002, pp. 2-3.
- [3] J. Mitola III, Software Radio Architecture. Object-Oriented Approaches to Wireless Systems Engineering. Canada: John Eiley & Sons, Inc, 2000, pp. 1-34.
- [4] M. E. Angoletta. (2003, Feb. 20). From Analog to Digital Domain. [Online]. Available: <http://humanresources.web.cern.ch/humanre->

- sources/external/training/special/ DISP2003/DISP-2003_L01A_20Feb03.pdf (20 Feb 2003).
- [5] A. K. Salkintzis, H. Nie, P. T. Mathiooulos. (1999, August). ADC and DSP Challenges in the Development of Software Radio Base Stations. *IEEE Personal Communications*.
 - [6] C. H. Dick. (2001). Design and Implementation of High-Performance FPGA Signal Processing Datapath for Software Define Radios. [Online]. Available: http://www.eetasia.com/ARTICLES/2001AUG/2001AUG09_ICD_AMD_RFD_TAC01.PDF.
 - [7] J. Huie, P. D'Antonio, R. Pelt. (2005). Synthesizing FPGA Cores for Software-Defined Radio. [Online]. Available: <http://www.altera.com/cp/fpga-cores-for-sdr.pdf>.
 - [8] S. Haykin, "Communication System," 4th ed. New York: John Wiley & Sons, Inc, 2004, pp. 362-365.
 - [9] H. Harada, R. Prasad, Simulation and Software Radio for Mobile Communications. Boston: Artech House, 2002, pp. 75-76.
 - [10] J. Mitola III, Software Radio Architecture. Object-Oriented Approaches to Wireless Systems Engineering. Canada: John Eiley & Sons, Inc, 2000, pp. 165-166.