

Cross Layer Transmission for AOS Packet Service

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Abstract—in this paper, the AOS packet service data transfers structure is presented first. Based on it, the effective throughput rate of AOS packet service is analyzed. Then, in order to improve the effective throughput rate, a scheme of AOS cross layer transmission system is presented. In the proposed scheme, based on the space channel state, the dynamic optimization parameters of network layer, Space Date Link Protocol sublayer and Sync and Channel Coding sublayer in AOS are selected. Then the switch thresholds are presented. Finally, the effective throughput rate of the proposed AOS cross layer transmission system is simulated. Simulations show that the throughput rate can be optimized effectively.

Index Terms—AOS, packet service, cross layer transmission, effective throughput rate

I. INTRODUCTION

AOS is used to manage and transfer various space mission data, such as the exploring and controlling data, the space communications data and other effectual loads data, over the space-to-ground, ground-to-space or space-to-space communication links. The packet service of AOS is used for the efficient transfer of various type packet data. It needs to offer high speed and reliability for various space mission packet data and afford the Quality of Service for end to end transfer.

However, the time variety character of space link will limit the system throughput performance of AOS. So it is important to conquer the effect of space link (or physical channel) for improving the AOS packet service effective throughput rate. The throughput performance of AOS

packet service has been studied by reference [1] which discusses the impacts of frame length and the relevant solutions on the throughput. Especially one adaptive technique with dynamic sizing of the Space Date Link Protocol sublayer frame explored. It shows that optimal sizing of the SDLP sublayer frame in the presence of varying channel noise indeed has a large impact on the user seen throughput. In addition, it shows that adaptive frame length control can be exploited to improve a desired level of throughput, and to extend the usable radio range with graceful throughput degradation. But it assumes the modulation is the same and only maximum transmission unit changes according to the space channel conditions to improve AOS throughput, range, and efficiency. And the cross layer transmission scheme has not proposed. However, in wireless systems, recent research has advocated many link layer techniques including error control, channel state dependent protocols and variable spreading gain and the cross layer techniques are generally used to enhance the throughput performance through adapting the different layers transmission parameters to the variations of the wireless channel. Nowadays, the concept of cross layer has been widely accepted. As a result, a large number of cross layer design proposals have appeared in the recent reference [2]-[8].

In this paper, a cross layer transmission scheme of AOS packet service is proposed to increase the AOS packet service effective throughput rate based on intrinsic relations between the channel state, the throughput rate and each parameter of AOS different layers. To detail the principle and performance of the proposed scheme, the remainder of the paper is

organized as follows: A brief explication of the AOS packet service data transfers structure and cross layer structure is given in section 2 and section 3 respectively. In section 4, AOS effective throughput rate formula of is introduced and analyzed. Then the AOS cross layer system is designed in section 5, followed by concluding remarks drawn in Section 6.

II. AOS DATA TRANSMISSION STRUCTURE

The AOS data transfers structure is layered. Comparing with OSI, AOS packet service structure is shown as Fig. 1.

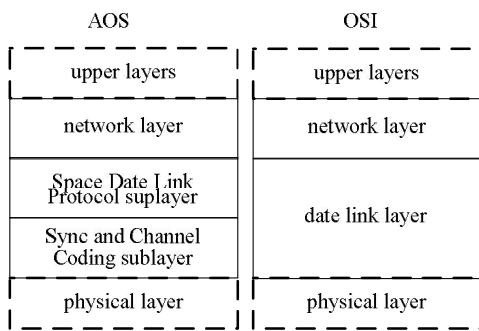


Figure 1. AOS packet service structure compare with OSI

The AOS standard advises that the continuous data from upper layers are partition to be variable length packets in network layer. The different packet service data transferring from network layer are partition to be fixed length transfer frames in Space Date Link Protocol (SDLC) sublayer. Then, these AOS transfer frames are coded and placed an Attached Sync Marker (ASM) in Sync and Channel Coding (SCC) sublayer, and send in physical layer. The traditional AOS layered structure is designed for the worst or the average space channel, and the communications in adjacent layers is invariable. It is unable to adapt the time-varying capacity and high bit error rate (BER) space channel. Consequently, increasing the throughput performance requires a cross layer joint design and optimization approach to dynamically adjusting the parameters of different layers.

III. AOS CROSS LAYER TRANSMISSION STRUCTURE

The cross layer transmission structure of AOS packet service is shown in Fig. 2. Obtaining the channel Signal Noise Ratio (SNR) from the estimator, controller decides the packet length of network layer, the frame length of

SDLP sublayer, and coding efficiency and frame sync parameters of SCC sublayer to maximize the AOS packet service effective throughput rate.

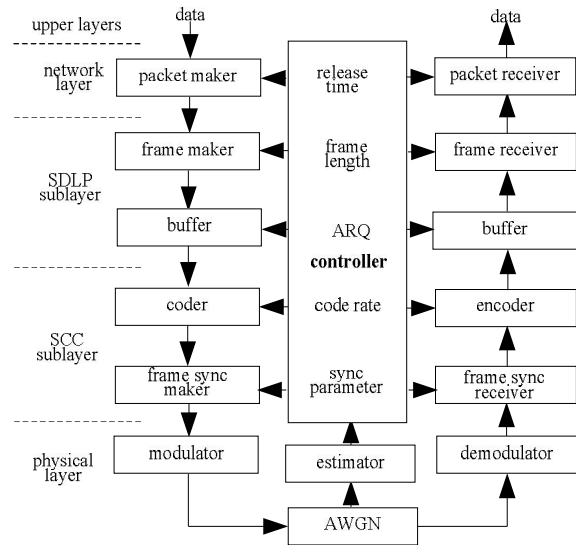


Figure 2. AOS cross layer structure

The AOS cross layer system is research with following presumptions and variable introductions:

1 SNR of current channel can be check and feed back to the sender accurately.

1 The frame queue in SDLP sublayer is saturated and the “bad” frames can be detect and drop reliably with CRC field.

1 $T_{effective}$ is the AOS effective throughput rate.

1 E_{up} is the efficiency of upper layers. It relates to service PRI and AOS multiplexing.

1 L_{frame} is the length of frame.

1 L_{packet} is the average packet size.

1 L_{packet_c} is the length of packet header and control field.

1 L_{frame_c} is the length of frame primary header and CRC field.

1 L_{frame_opt} is the optimized length of frame.

1 L_{frame_ada} is the adaptive length of frame.

1 L_{FLM} is the length of Frame Length Marker.

1 E_{frame} is the efficiency of frame.

1 R_{code} is code rate.

1 E_{sync} is the efficiency of frame sync.

1 L_{ASM} is the length of ASM.

1 $R_{modulation}$ is modulation rate.

1 P_{sync} is the successful probability of frame sync.

1 $P_{success_frame}$ is the successful probability of receiving frame.

1 R is the symbol rate of physical channel.

IV. ANALYSIS OF AOS EFFECTIVE THROUGHPUT RATE

The effective throughput rate is the ratio of successfully received bit numbers to the total sending bit numbers. Based on the section 3, we have:

$$T_{effective} = E_{up} E_{packet} E_{frame} R_{code} E_{sync} R_{modulation} \cdot P_{success_sync} P_{success_frame} R \quad (1)$$

Where,

$$E_{packet} = \frac{L_{packet} - L_{packet_c}}{L_{packet}} \quad (2)$$

$$E_{frame} = \frac{L_{frame} - L_{frame_c}}{L_{frame}} \quad (3)$$

$$E_{sync} = \frac{L_{frame}}{L_{frame} + L_{ASM}} \quad (4)$$

Obviously, the parameters of each layer affect the AOS effective throughput rate. So combined optimization can enhance the effective throughput rate.

V. DESIGN OF AOS CROSS LAYER TRANSMISSION

The AOS packet service transfers a sequence of variable-length, delimited, octet-aligned service packets which must have a Packet Version Number (PVN) authorized by CCSDS. So in AOS packet service, the packet processing, Virtual Channel (VC) generation, VC multiplexing, the frame framework, and sync and channel coding affect AOS effective throughput rate mainly.

A. Design of Packet Release Time in Network Layer

A sequence of variable-length packets are transferred from different data sources. The size of each packet is an exponential distributed variable, i.e.

$$f(l) = \frac{1}{L_{packet}} \exp\left(-\frac{l}{L_{packet}}\right) \quad (5)$$

Then packets are multiplexed to be a Multiplexing Protocol Data Unit (M_PDU) and transferred on a VC transfer frame. During a fixed mission phase, the length of M_PDU packet zone L_{M_PDU} is fixed and we assume that N is the maximal number of available packets which are multiplexed in a M_PDU packet zone at release time T_r . If there are insufficient packets arrived at T_r , the idle packets will be contained in the M_PDU packet zone. And if the number of packets arrived exceeds N at T_r , packets need to be split to fill the M_PDU packet zone completely, and start a new M_PDU on the same VC with the remainder. In this paper, the units of packet size and frame length are octet.

Letting $P(n)$ be the probability of n packets has arrived at T_r . The process of packets arriving can be modeled as a Poisson process with the average arriving rate ρ one per millisecond, then we can get $P(n)$ as

$$P(n) = \frac{(\lambda_p T_r)^n \exp(-\lambda_p T_r)}{n!} \quad (6)$$

So the efficiency of M_PDU E_{M_PDU} can be express as

$$E_{M_PDU} = \frac{\sum_{n=0}^N n \int_0^{\infty} l f(l) dl P(n) + \sum_{n=N+1}^{\infty} N L_p P(n)}{L_{M_PDU}} = \left(\sum_{n=1}^N (\lambda_p T_r)^n / (n-1)! + N \sum_{n=N+1}^{\infty} (\lambda_p T_r)^n / n! \right) \cdot \frac{L_{packet} \exp(-\lambda_p T_r)}{L_{M_PDU}} \quad (7)$$

Assuming $L_{packet}=100$ bytes, $L_{M_PDU}=1000$ octets, $N=10$, the relations of E_{M_PDU} to T_r with different values of ρ are shown as Fig.3.

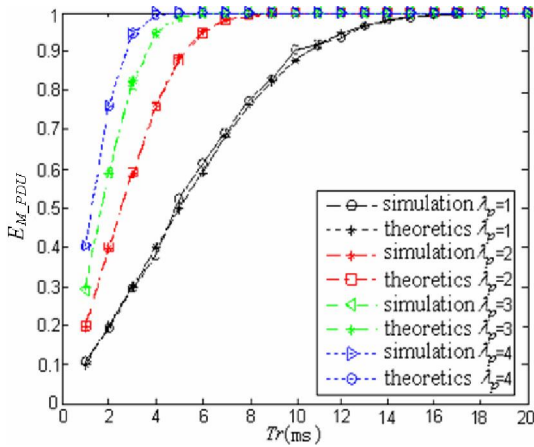


Figure 3. the relation of E_{M_PDU} and T_r

From Fig.3, we can see that the simulation results are close to the theoretic results. So the reliability of (7) is proved. We also know that if T_r is long enough, E_{M_PDU} will trend to the maximum value 1 and the larger λ_p is, the shorter T_r for a M_PDU will be. So we assume the T_r is long enough.

B. Design of Frame Length in SDLP Sublayer

If the effect of SCC sublayer and the repeat transfer frame are not considered, the frame error rate FER can be expressed as

$$FER = 1 - (1 - BER)^{8L_{frame}} \quad (8)$$

So we can get $P_{success_frame}$ as

$$P_{success_frame} = 1 - FER = (1 - BER)^{8L_{frame}} \quad (9)$$

Substituting (2) ~ (4) and (9) into it, let (1) satisfies (10).

$$\frac{\partial \ln T_{effective}}{\partial L_{frame}} = 0 \quad (10)$$

Then we can get L_{frame_opt} maximizing the throughput as

$$L_{frame_opt} = \sqrt{\frac{(L_{frame_c} + L_{ASM})^2 - (L_{frame_c} + L_{ASM})}{8 \ln(1 - ber)}}$$

$$+(L_{frame_c} - L_{ASM}) \quad (11)$$

From (11), we can obtain that L_{frame_opt} is only determined by BER .

Based on the analysis above, we present a frame length adaptive scheme. Before sending data, a pretreatment which chooses an appropriate frame length according to the channel states. The pretreatment can be performed by software, which needn't change the AOS architecture, so it is can be realized easily. Apparently, L_{frame_opt} is continuous in (11). However, continuous frame length may increase the system complexity and cause performance instability. The discrete frame lengths are more suitable for the practical data transmission, and the frame length being an integer times of byte can be treated easily. As Table 1, the discrete frame lengths 128, 312 and 1024 bytes are considered and the Frame Length Marker (FLM) is defined. We define the FLM in frame primary header as Fig.4. So, $L_{FLM}=2\text{bits}$.

TABLE I. SELECTION OF FRAME LENGTH AND FLM

BER	$L_{frame_ada}(\text{bytes})$	FLM
$BER < 6 \times 10^{-6}$	1024	00
$6 \times 10^{-6} < BER < 2 \times 10^{-5}$	512	01
$2 \times 10^{-5} < BER$	128	10

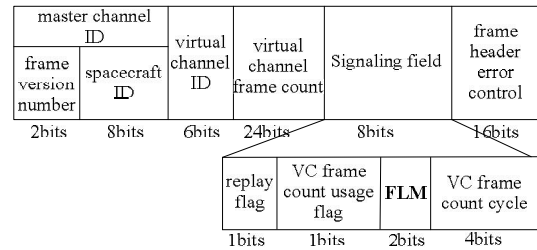


Figure 4. design of frame primary header

Letting $L_{frame_c}=12\text{bytes}$, $L_{ASM}=8\text{bytes}$, and $R=1\text{symbols/s}$, $T_{effective}$ with L_{frame_opt} and L_{frame_ada} is shown in Fig. 5. When $BER > 10^{-4}$, the communications will be stopped because of too low $T_{effective}$. Fig. 5 shows that $T_{effective}$ with L_{frame_ada} is close to which with L_{frame_opt} . So if channel states change, $T_{effective}$ can be improved effectively by the proposed adaptive frame length

scheme compared with fixed frame length scheme, and the switch frequency of frame length is not as high as the optimal length scheme.

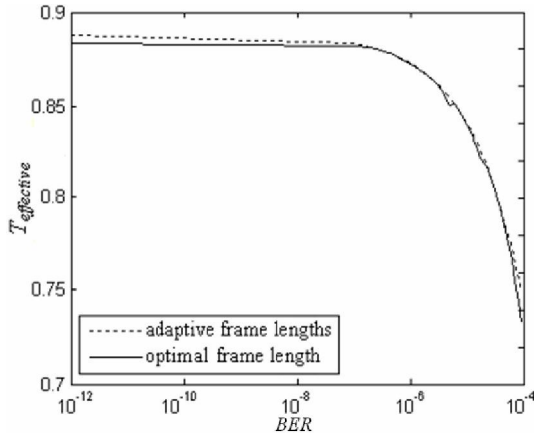


Figure 5. $T_{effective}$ with optimal and adaptive frame lengths

C. Design of ARQ in SDLP Sublayer

Usually, space networking has to deal with hazardous communication conditions, such as long propagation delay, intermittent link connectivity, and high BER. In this environment, the employment of conventional automatic repeat request (ARQ) scheme is encouraged and specific protocols are preferred in order to assure the reliability of the data communication. In this view, a possible choice is represented which implements an enhanced ARQ scheme based on negative acknowledgments (NAK).

Setting the maximum times of every frame transmission is K_{max} , the efficiency of ARQ can be get as

$$Eff_{ARQ} = \frac{1}{\left[\sum_{k=1}^{K_{max}-1} k(FER)^{k-1}(1-FER) + K_{max}FER^{K_{max}-1} \right]}$$

$$= \frac{1+FER}{(1-FER)^{K_{max}}} \quad (12)$$

However, the frame error ratio after ARQ is

$$FER_{ARQ} = FER^{K_{max}} \quad (13)$$

So we can get $P_{success_frame}$ as

$$P_{success_frame} = 1 - FER^{K_{max}} \quad (14)$$

D. Design of Code in SCC Sublayer

Low Density Parity Check (LDPC) code is a class of block codes with near Shannon limit performance. It can provide very low error floors and very fast iterative convergence. These qualities make LDPC code a good fit for near Earth applications where very high data rates and high reliability are the driving requirements^[9]. And LDPC code has been submitted to the CCSDS channel coding in the fall 2002 Houston meeting. In this paper, the linear block code is designated by (L_{frame}, R_{code}) . Fig. 6 shows the FER (solid) and the simulation FER (dashed) for the LDPC codes by BPSK modulation in physical layer. From left to right, these six codes have parameters (128, 1/2), (512, 2/3) and (1024, 4/5).

The simulation FER can be implemented as (15), and the simulation parameters are shown as Table II.

$$FER = p \exp(-q \cdot SNR) \quad (15)$$

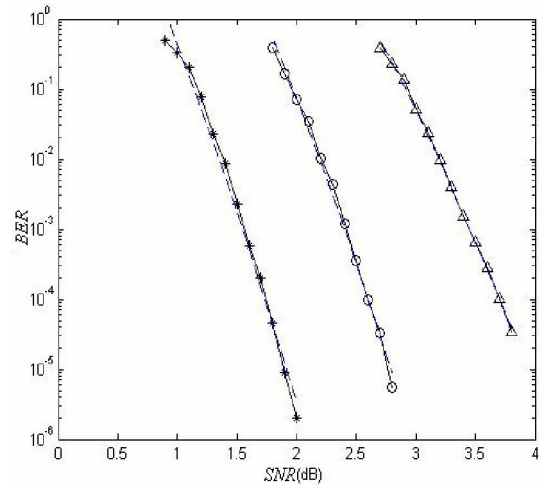


Figure 6. FER for the LDPC codes

TABLE II.

THE SIMULATION PARAMETERS FOR FER

model	L_{frame_ada}	R_{code}	p	q
1	128	1/2	109.56	21.55
2	512	2/3	10.12	2.23
3	1024	4/5	9.81	1.87

E. Design of frame sync in SCC Sublayer

Frame sync is achieved by sync of an ASM associated with each LDPC Code block. The ASM is a bit pattern of CCSDS Recommended Standard^[10], and it precedes the

LDPC codeblock. The searching and checking process of frame sync as Fig.7.

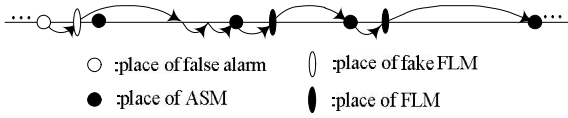


Figure 7. searching and checking process of frame sync

Frame synchronizers should be set to expect a marker at a recurrence interval equal to the length of the ASM plus that of the LDPC Code block. All code models use the ASM as Table III [10, 11].

TABLE III.
THE BEST CODE MODEL OF ASM

ASM code model	L_{ASM} (bits)
EB90	16
1ACFC1D	32
FFF2D58B65466000	64

Setting a limit to the frame sync error, the probability of false alarm P_{FA} and the probability of false leak P_{FL} are expressed as

$$P_{FA} = \sum_{i=0}^J C_{8L_{ASM}}^i (1/2)^{8L_{ASM}} \quad (16)$$

$$P_{FL} = 1 - (BER)^{L_{FLM}} \sum_{i=0}^J C_{8L_{ASM}}^i (1 - BER)^{8L_{ASM} - i} BER^i \quad (17)$$

Where, J is the maximal number of error bits in ASM.

The simulation of P_{FA} and P_{FL} with different ASM code models and J are shown in Fig.8 and Fig.9.

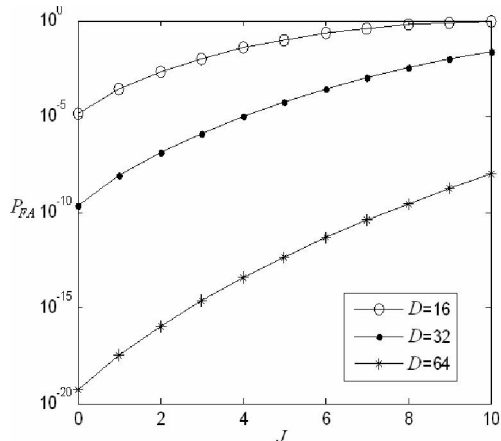


Figure 8. P_{FA} with different ASM code models and J

From Fig.8 we can get that the value of P_{FA} becomes large along with the increasing value of J and becomes small along with the increasing value of L_{ASM} . From Fig.9 we can get that the value of P_{FL} becomes large along with the increasing value of BER and becomes small along with the increasing value of J . So we can get the optimum values of L_{ASM} and J as Table IV.

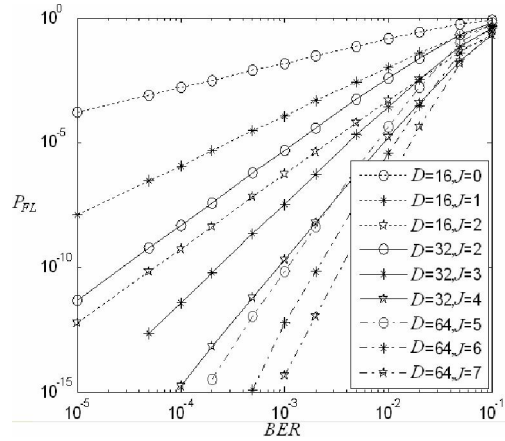


Figure 9. P_{FL} with different ASM code models and J

TABLE IV.

THE OPTIMUM VALUES OF L_{ASM} AND J

L_{ASM} (bits)	J (bits)
16	1
32	3
64	7

Setting a is the times for checking the ASM successfully, the probability of false sync P_{FS} can be defined as

$$P_{FS} = (1 - (1 - P_{FA})^{L_{frame}/2}) P_{FA}^{a-1} \quad (18)$$

Setting $L_{frame}=512$ bytes and the value of J as the Table IV, P_{FS} can be gotten as Table V.

TABLE V.
THE PROBABILITY OF FALSE SYNC

L_{ASM} (bits)	a (bits)			
	1	2	3	4
16	2.0×10^{-4}	5.9×10^{-8}	1.1×10^{-11}	3.0×10^{-15}
32	6.7×10^{-9}	8.5×10^{-15}	1.1×10^{-20}	1.4×10^{-26}
64	6.0×10^{-18}	2.3×10^{-28}	8.7×10^{-39}	3.3×10^{-49}

From (16) and Table V, we can get if the value of J is appropriate, P_{FA} and P_{FS} can be ignored. So, P_{sync} can be defined as

$$P_{sync} = \frac{(1 - P_{FL})^a}{(1 - BER)^{L_{FLM}}} \quad (19)$$

The relations of P_{sync} with different a and ASM code models are shown in Fig. 10, Fig. 11 and Fig. 12.

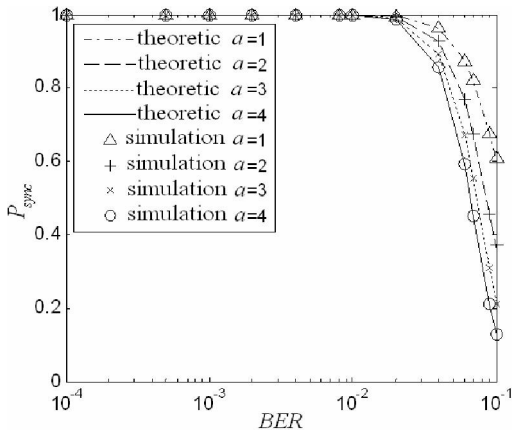


Figure 10. P_{sync} with different a and $L_{ASM} = 16$ bits

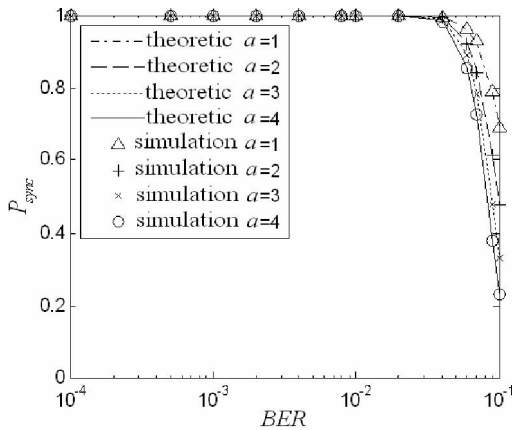


Figure 11. P_{sync} with different a and $L_{ASM} = 32$ bits

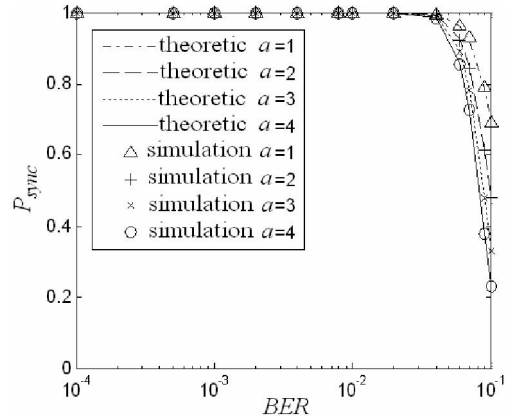


Figure 12. P_{sync} with different a and $L_{ASM} = 64$ bits

From Fig.10, Fig.11 and Fig.12, we can see that the simulation results are close to the theoretic results. So the reliability of (19) is proved.

Consequently, the parameters of AOS frame sync can be get as Table VI. Hereinto, the effect of frame sync is best at $L_{ASM} = 64$ bits, $J = 7$ bits and $a = 1$ bits.

The relation of channel SNR to BER are given as follows

$$BER = erfc(\sqrt{SNR}) / 2 \quad (20)$$

TABLE VI.
THE PARAMETERS OF AOS FRAME SYNC

ber	L_{ASM}	J	a
$ber \leq 10^{-4}$	16	1	≥ 4
$10^{-2} \leq ber < 10^{-4}$	32	3	2or3
$10^{-2} < ber$	64	7	1

Then

$$P_{sync} = \frac{(1 - P_{FL})^a}{(1 - BER)^{L_{FLM}}} = \frac{(1 - P_{FL})^a}{(1 - erfc(\sqrt{SNR}) / 2)^{L_{FLM}}} \quad (21)$$

$P_{success_frame}$ is given as

$$P_{success_frame} = P_{sync} (1 - p \exp(-q \cdot SNR)) \quad (22)$$

Setting $E_{up} R_{modulation} R = k$, $T_{effective}$ is given as

$$T_{effective} = k(L_{frame} - L_{frame,c})(1 - (\text{pexp}(-q \cdot \text{SNR}))^{K_{max}+1}) \left(\frac{\text{erfc}(\sqrt{\text{SNR}})}{2}\right)^{L_{FLM}}$$

$$R_{code} \frac{\left(\sum_{i=0}^J C_{8L_{ASM}}^i \left(1 - \frac{\text{erfc}(\sqrt{\text{SNR}})}{2}\right)^{8L_{ASM}-i} \left(\frac{\text{erfc}(\sqrt{\text{SNR}})}{2}\right)^i\right)^a}{(L_{frame} + L_{ASM})(1 - \text{erfc}(\sqrt{\text{SNR}})/2)^{L_{FLM}}} \quad (23)$$

F. Design of switch thresholds of cross layer system

The AOS cross layer system is simulated with following presumptions: $L_{ASM}=64\text{bits}$, $J=7\text{bits}$, $a=1\text{bits}$, $K_{max}=1$, $k=1$ and other parameters as above paragraphs. The simulation of $T_{effective}$ with different models is shown in Fig.13.

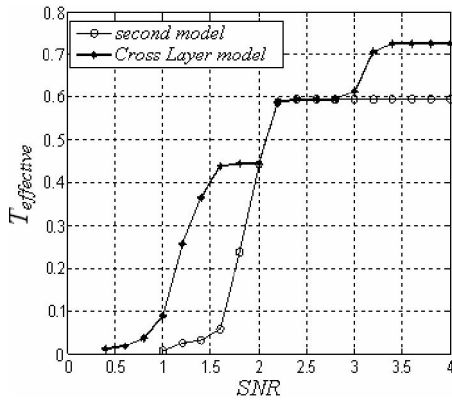


Figure 13. $T_{effective}$ with different models

From Fig.13 we can get the switch thresholds of SNR as Table VII. Here the cross layer scheme is proposed to maximize throughput by suitably determining discrete adaptive frame length L_{frame_ada} based on different ranges of SNR. When $\text{SNR} < 1.5$, the communications will be stopped because of too low $T_{effective}$.

TABLE VII.

THE THRESHOLD OF AOS CROSS LAYER SYSTEM

model	SNR threshold(dB)
1	1.80
2	2.01
3	2.98

Fig. 13 shows the effective throughput rate of AOS cross layer scheme. So if channel SNR change, the throughput rate can be improved effectively by the proposed scheme compared with three fixed schemes, and the switch frequency of cross layer system is not

high. The scheme is simple and need not change the frame length and code rate frequently, so the stability of the AOS and the reliability of data transmission can be guaranteed. So it is more suitable for practical data transmission.

VI. CONCLUSIONS

An AOS cross layer scheme uses appropriate packet release time, frame length, code rate and sync parameter for the throughput rate improvement is presented in this paper. First of all, an effective throughput rate formula of AOS has been proposed, from which we know that the effective throughput is mainly affected by packet release time, frame length, coding and channel SNR. So a cross layer scheme controlling these parameters of different layers based on the ranges of SNR is provided for maximizing the effective throughput rate, which needn't modify the AOS protocol too much. Simulation results show that compared with fixed transmission scheme, the proposed scheme can improve the AOS throughput rate effectively. Using the presented result, AOS protocol designers can verify and optimize their designs which provide a reference to the future application of space missions. This research does not consider the probability of modulation to AOS effective throughput rate. Further research is needed to support the performance analysis of more complicated communications. In addition, it was proven that the proposed AOS cross layer scheme provides an effective alternative to achieve high throughput under AWGN. The performance of the proposed AOS cross layer scheme in fading condition is future work.

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