

The Performance Analysis of Multi-Layer Split-Protocol

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Abstract

The growing need for high-performance networking is achieved with parallel processing; several networking functions are processed concurrently in order to accomplish a performance Networking Architecture. Open systems interconnection (OSI) model is an example of multi-layering structure, and each layer performs definite function unique to that layer. OSI model works on pass it on principle, and it is divided in two stacks lower stack and upper stack. Layers 4 - 7 represent upper stack and responsible for data applications. The remaining 1 - 3 layers represent the lower stack and mostly involve in data movement. There are many techniques are available for server optimization enhancing the availability by distributing the load among peer servers. According to our knowledge, nobody has implemented such splitting architecture across the entire OSI model. In this paper, we present multilayer Split-protocol (MLSP) a high performance, reliable and secure technique for spiting an application or network protocol across OSI model, and we present the design, implementation, and empirical performance evaluation of MLSP. It is the ideal choice for Cloud services where each functional component is considered an independent of each other.

Keywords

OSI Model, Bare Machine Computing, Split-Protocol, M-TCP, Web Server, Reliability, Availability, Performance

1. Introduction

Web server security and load delivery between Web servers are significant challenges that remain to be resolved using a range of methods. In specific, load-balancing and splitting techniques are employed at numerous layers of the protocols tack to segment the load among a cluster of Web servers [1] [2]. A P2P application such as Bitterant (or anything else named "torrent"), Gnutella, Lime Wire, and KaZaA are some examples of P2P applications. P2P applications work by letting your computer down load parts of files from multiple sources on the Internet at the same time [3]. We propose a technique for splitting web request at various layers. At upper layers (5 - 7) an application data and bottom layers (1 - 4) mostly moving data around. HTTP/TCP connection between Web servers permits one server to manage a connection establishment and others to accomplish data transfer. We are splitting TCP connections in such way that it allows servers to distribute the load without the central authority or any client's help. Each server has multiple NIC cards. These NIC cards can function unidirectional or bi-directionally based on a server's algorithm. Details of different type's HTTP splitting into two servers are explained in [4] [5]. Figures 1-3 describes splitting in to multiple servers with a different configuration. Split-protocol is used for improving security, availability, and reliability of server system [6] [7]. Jaliya and Fox have described several different studies of clouds and cloud technologies on both real applications and standard benchmark. They found that cloud technologies work well for most pleasingly-parallel problems ("Map-only" and "Map-reduce" classes of applications) [8].

Novelty and our original contribution: According to our knowledge we have not found any existing technique which splits a single web service on multiple servers, and each server performs the function of an individual or multiple OSI layers. The OSI reference model illustrates the layered architecture, at every layer specific actions, functions, and services take place. We can divide network functionality into seven subprocesses and implement similar to the OSI model. The multi-layer splitting offers an efficient load balancing; minimizing the dependency; improves the reliability by distributing the functionality over numerous servers; reduces the system complexity and improves performance and the system security.









Configuration 1

Configuration 2

DS: Data Server



CS: Connection Server



Figure 3. Splitting between two NIC cards.

The remainder of the paper is organized as follows. Section II exhibits related work and Section III talks about some performance enhancing techniques. IV describes multilayer splitting architecture. Section V presents M/M/1 and M/M/S Queuing systems. Section VI outlines the MLSP implementation, and Section VII defines performance metrics. Section VIII presents performance. Section IX details the signification of MLSP. And section X highlights an acknowledgment. Section XI holds the conclusion.

2. Related Work

Bare Machine Computing (BMC), formerly stated to as dispersed OS computing [9], permits an application to runonabare PC without any centralized kernel or OS support. In the BMC methodology, an application condensed in application object (AO) [9] straight talks with the hardware. According to my knowledge, a technique form ultilayer splitting, a HTTP-based TCP connection in this custom

has not been introduced earlier. Splitting is comparable to the M-TCP protocol to transfer TCP connections from one server to another [10]. TCP connection splitting and the in [11], staticanalys is techniques are used to analyze the code of Linux device drivers. Also, the communication between controllers, the OS, and the hard ware is monitored. In [12], hard ware abstraction and APIs for devices and software interfaces are employed as a foundation for producing device drivers [13]. The Uniform Driver Interface (UDI) is designed for generating device drivers that are manageable concerning platforms and OSs [14]. The Network Driver Interface Specification (NDIS) separates the NIC from its driver and defines a standard interface between upper and lower layer controllers [15]. The NDIS library is supported by many Windows versions, while the NDIS wrapper [16] enables Windows drivers to be used with Linux. [17] uses TCP slicing for performance guarantee for cluster-based internet services. In [18], an approach for driver reuse is proposed, where a virtual machine is used to run a driver with its original OS. Various methods applied for load balancing share some essential features. For example, both load balancing and TCP connection splitting enable a server load to be allocated between servers. In layer-2 (link layer) load balancing, various links function as a unique virtual connection, and in layer-4 (transport layer) load balancing, applications are processed based on port numbers and IP addresses. In layer-7 load balancing [17], content knowledge from the HTTP request is accepted to perform load-balancing judgments. In this case, there are two TCP connections that are spliced [18]. G. Anastasi [19] examined the BitTorrent scheme and completed the preliminary study. Gummadi et al. [20] investigated the prevalence of P2P content across the Internet and described the "download at most once" feature of P2P clients. Saroiu [21] [22] characterized the P2P traffic over the Internet, including Napster, Gnutella, and KaZaa systems. Izal [21] investigated a five-month workload of an individual BitTorrent method for software delivery that required thousands of peers. Bellissimo and Izal each estimated the realization of BitTorrent at the flash crowd period [22] [23]. The security aspect of the splitting technique is demonstrated in earlier publications [7] [24] [25]. Most of the existing related work splitting needs some shorts of client involvements and requires some central controlling mechanism for communicating between peers. Existing techniques heavy weight communication overhead affects the performance of overall systems. In our technique there is only two packets of 168 byte are involved for inter server communication and it does not require any client's involvement. Spiting process is entirely transparent to client.

3. Performance Enhancing Techniques

3.1. Pipeline

The pipeline is an execution procedure where several tasks are an overlay in execution. The computer pipeline is distributed in various stages. In each step, complete a portion of execution in parallel. Each level is connected to next to create a pipe. Typically, pipeline does not reduce the proper time execution; instead, it improves performance, throughput. A computer processor pipeline is from time to time shared with an instruction pipeline and an arithmetic pipeline. The instruction pipeline signifies the phases in which a step is progressed through the processor, including it is being drawn, possibly buffered, and then executed. Moreover, the arithmetic pipeline embodied the portions of a mathematic operation that can be broken down and overlapped as they are performed [18]. A similar technique used in dynamic random-access memory (DRAM), in which the memory loads the required memory subjects into a small cache composed of static random-access memory (SRAM) and then immediately begins fetching the next memory contents. This makes a two-stage pipeline, wherein read instruction is one stage written to memory is in the other phase [14]. The Uniform Driver Interface (UDI) is designed for generating device drivers that are manageable concerning platforms and OSs [6].

3.2. Parallel Computing & Distributed Computing

Distributed computing and parallel computing are frequently deliberated similar, though there is a trivial difference between them. Parallel computing denotes a tightly coupled system with each other. If we look closely at parallel computing, it relates to the technique in which computation is divided between various processors who are sharing the collective memory. The uniformity describes the design of parallel computing. The cluster of nodes connected through an InfiniB and network which is configured with shared memory. Distributed computing, on the other hand, comprises any design or scheme in that computation is fragmented into parts and are performed simultaneously on unalike computing components, they may be multiprocessors on the same node, another computer or cores within the same CPU [26] [27].

4. Multilayer Split Architecture

For readers convenience we are reproducing the splitting architecture between two or more servers [4]. The split architecture used for the experiments described in this paper is illustrated in **Figure 2**. Although these experiments were conducted in a LAN environment, as noted earlier, the proposed splitting technique does not require that the set of clients {C} be connected to a LAN (they can be located anywhere on the Internet). The only requirement is that the servers be connected to the same LAN for the reasons discussed below. However, this requirement does not limit the scope or scalability of splitting since many real-world Web server clusters are located within the same LAN. The clients send requests to servers S1 or S2. S1 and S2 are referred to as split servers. For a given request, the connection server (CS) handles the {CE, CT} phases of a connection, and its delegated server S2 (DS) handles the {DT} phase. Similarly, S2 can act as a server for a client's request and its DS will be S1. The clients do not have any knowledge of a DS. A given request can also be processed by the CS without using the DS. In general, there can be a set of n (≥ 2) servers that can delegate requests to each other.

A given request is split at the GET command as shown in **Figure 1**. The CS handles the connections, and the DS handles the data transfer. In addition to connections, the CS also handles the data ACKs and the connection closing. The CS has complete knowledge of the requested file, its name, size, and other attributes, but it may or may not have the file itself. However, the DS has the file and serves the data to the client. When a TCP connection is split in this manner, the TCP sliding window information is updated by S1 based on received ACKs even though the data file is sent by S2. Likewise, S2 knows what data has been sent, but it lacks knowledge of what data has been actually received by the client. Thus, retransmissions require that ACK information be forwarded by S1 to S2 using delegate messages as described below. The number of delegate messages exchanged should be kept to a minimum since they add overhead to the system and degrade performance.

When a client makes a request to S1, its connection is based on (IP3, Source-Port) (IP1, 80). S1 can serve this request to a client directly, or it can utilize its DS, which is S2, to serve data. The decision to use a DS can be made based on several factors such as the maximum number of requests that can be processed at S1, the maximum CPU utilization at S1, or resource file location. Alternatively, a load balancing algorithm could be used. When S1 chooses to use S2 as a DS, it proceeds as follows. After the GET command is received, it sends an ACK to the client and also sends a delegate message DM1 to S2 (e.g. DM1). The message DM1 contains the state of the request that is stored in S1 in the form of an entry in the TCP table (referred to as a TCB entry). When DM1 reaches the DS, it creates its own TCB entry and starts processing this request as if it was initiated in the DS itself. When a DS sends data to the client it uses the CS's IP (IP1).

In principle, the Internet setting is not different from a LAN environment since the DS does not need to receive any packets sent by the client to IP address IP1. A client located anywhere on the Internet can communicate in the usual manner with the CS. Since it is unaware that the DS is actually sending the data, it sends the ACKs as usual to the CS with whom the TCP connection was established. From the client's point of view, it has established a connection with IP address IP1. Now consider the information that is present in the local routers and switches assuming that both S1 and S2 are located in the same LAN. Note that only S1 should respond to ARP requests for IP address IP1. This ensures that any local router receiving the response will have a single ARP entry (IP1, MAC S1) in its ARP cache and correctly forward ACKs and connection requests sent with destination address IP address IP1 to S1. Note also that the switch to which S1 is connected will have the entry (MAC S1, INT1) in its forwarding table, where INT1 is S1's interface to the switch. Likewise, the switch to which S2 is connected has the entry (MAC S2, INT2) in its forwarding table, where INT2 is S2's interface to the switch. When S1 sends a delegate message to S2, if they

are both on the same LAN, S1 can simply encapsulate the message in a MAC layer frame addressed to MAC S2 (*i.e.*, it does not need an IP address to receive delegate messages from S1). Thus, with these assumptions, switches and routers do not need any special configuration for split connections to work.

However, if S1 and S2 are on LANs with different subnet prefixes (or in general, on WANs or different networks) and communicate through routers, S2 is not reachable using IP address IP1 since its prefix is not consistent with the network it is on. So, it will need to use its own IP address IP2 to receive packets including delegate messages from S1. This means that the router for S2 must have an ARP entry (IP2, MAC S2) for forwarding to S2, which will only be present if S2 has responded to ARP request for IP address IP2 with its MAC S2. But in this case, if S2 is also sending data to a client using IP address IP1 as source, it raises a security issue due to IP address spoofing. Such IP address spoofing may cause problems with firewalls due to sending topologically incorrect packets. For splitting to work in this case, note that S1 must send and receive with IP address IP1, whereas S2 must send with IP address S1 and receive with IP address S2. Now S1 and S2 cannot both delegate to each other, since it is not possible for both machines to send and receive using both IP addresses IP1 and IP2. There are also TCP issues with splitting due to its sliding window, duplicate acks, fast retransmit, and congestion control that need further study. More delegate messages could be used to address some of these TCP issues, but this would have a negative performance impact.

As the connection and data transfer are split in the architecture, there is a need to send one or more DM messages (DM2s) to DS. At least one DM2 message is needed to indicate that CS received the FIN-ACK. If a received ACK indicates that data is lost, retransmission is needed. One or more DM2s are needed to handle retransmissions since the DS does not receive any data ACKs. The CS monitors the data ACKs and decides to send DM2s as needed. Throughout the splitting process, the client is not aware of DS, and there is no need to involve the client (unlike M-TCP). The last DM2 message to DS is used to terminate when all data has been acknowledged by the client.

Splitting results in two possible overheads. Network traffic due to sending DMs to DS, and the latency encountered at the client due to DM transmission on the LAN (or WAN) from CS to DS. In a LAN environment, this latency is negligible, but may be larger in a WAN or Internet environment. The network traffic generated for each request is at least two DM packets; in most cases it is two packets assuming no retransmissions. If the DM packet is small (168 bytes in a bare PC), the network overhead will be reduced. However, one needs to consider the above two overheads of the split request architecture for a given application.

The MLSP design used for the experiment is exemplified in Figures 1-3. Although the experimental tests were conducted in a LAN/WAN environment, the detailed splitting architecture is described in our earlier works [4] [5]. The clients do not require any prior knowledge of DSs. The CS can also process a given request without using the DS. In general, there can be a set of n (≥ 2) servers those can delegate requests to each other.

When a client requests CS, its connection is based on (IP, Source Port) (IP1, 80). S1 can serve this request to a client directly, or it can utilize its DS, which is DS1, to serve data. The decision to use a DS can be made based on several factors such as the maximum number of requests that can be administered at CS, the maximum CPU utilization at CS, or resource file location. Alternatively, a load-balancing algorithm could be used. When CS chooses to use DS, it proceeds as follows. After the GET command is received, it sends an ACK to the client and sends a delegate message (DM1) to DS. The message DM1 encompasses the state of the request that is kept in CS as TCB records. When DM1 reaches the DS, it creates its TCB entry and starts processing this application as if it was initiated in DS itself. When a DS directs data to the client, it uses the CS's IP (IP1).

Figure 2 represents two split configurations 1 & 2 with only one NIC card. In configuration 1, DSs just sends data packets to the clients and does not send any other kinds of packets to the clients.

In configuration 2, the DSs dose sends other types of packets to clients. **Figure 3** represents two split configurations 3 & 4 with two NIC cards. In configuration 3, DSs only sends data packets to the clients and does not send any other types of packets to the clients. CS and DSs both receive all packets through NC1 and transmit through NC2. In configuration 4, DSs do receive/send other types of packets to the clients. Also, CS and DSs both receive all packets through NC1 and transmit through NC2 respectively [28].

Figure 4 represents partial delegation CS and DS both can send data packets and NIC1 & NIC2 both can receive and transmit data (full duplex).





5. M/M/1 and M/M/S Queuing System

As shown in **Figure 5**, MLSP system with a single queue with more than one parallel server, then we have M/M/S queuing system, in the diagram below four identical servers severing a request coming into a single queue [19].

The arrival rate λ , Service rate is μ , the ratio of traffic intensity

L

$$P = \lambda/\mu \tag{1}$$

$$Lq = Ls - LB \tag{2}$$

Lq is average number costumer in the waiting line for services, Wq is average time customer spend in the waiting line for services

$$Vs = Wq + (1/\mu) \tag{3}$$

We are reproducing hear the analytical model of split-system as shown in **Figures 5-8**, represents a Non-split system. It has to perform both tasks of CS and DS (establishing a connection and serving data) [29]; which is similar to single-channel, multiphase system. For one http request, the system is locked for entire time t and can offer μ = mean service rate. A split system become sidle after t/2 time and offers 2 μ mean service rate.

Figure 7 shows the non-resource sharing systems non-split system and **Figure 7**. Illustrate the resource sharing non-split system and Multichannel, single-phase system.

Figure 8 demonstrates split system two hardware systems and each hardwares has two software components CS and DS. This system is similar to Multichannel, multiphase system.

We will consider the case that the server can serve limited http requests. The waiting queue is regarded as a finite queue.

Where λ = mean arrival rate, μ = mean service rate, and n = the number of request in the waiting line system and the probability is described in (1).



Figure 5. M/M/s queuing system.



Non-Split system; Single-channel, multiphase system



Split system; Single-channel, single system

Figure 6. Non-split and split systems.



Figure 7. Non-recourse sharing non-split system.





$$\Pi_0 = \left(1 - \lambda/\mu\right) / \left(1 - \lambda/\mu\right)^{(M+1)}$$

 $\Pi_n = \Pi_0 [\lambda/\mu]^n$ where $n \le M$

$$E[N] = \left[\left(\lambda/\mu \right) / \left(1 - \lambda/\mu \right)_{-(M+1)} \lambda/\mu \right]^{(M+1)} / \left[1 - \left(\lambda/\mu \right)^{(M+1)} \right]$$
(4)

 Π_n is the probability of n http request being in the server system. Let *M* is the maximum requests in the system, and Π_M is a value Π_n when n = M.

We can define the probability that a http request (*job*) not join the system [28] in (2).

$$E[Nq] = E[N] - \lambda(1-\pi)M/\mu$$
$$E[R] = E[N]/\lambda(1-\pi M)$$
$$E[Rq] = E[R] - 1/\mu$$
(5)

We will compare the average response E[Rs] shown in (3), (4) and (5) and E[Rc] time described (6) and (7) by for the separate and non-split queuing systems respectively as illustrated in **Figure 7** and **Figure 8**. The first system corresponds to two independents M/M/1 queues, with $\rho = \lambda/2\mu$ [29].

$$E[Rs] = (1/\mu)/(1 - \lambda/2\mu) = 2/(2\mu - \lambda)$$
(6)

On the other hand, the shared queue sharing system in **Figure 10** is represented by M/M/2 system. To obtain E[Rc] first, we will calculate E[Nc].

$$E[Nc] = 2\rho + \rho (2\rho)^{2} / 2! - \pi_{0} / (1-\rho)^{2}$$
(7)

where
$$\pi_0 = (1 - \rho)/(1 + \rho)$$
 (8)

$$E[Nc] = 2\rho / (1 - \rho^{2})$$
$$[Rc] = E[Nc] = 1 / (\mu (1 - \rho^{2})) = 4\mu / (4\mu^{2} - \lambda^{2})$$
(9)

We have

E

$$E[Rs] = 2/(2\mu - \lambda) = 4\mu + 2\lambda/(4\mu^2 - \lambda^2) > E[Rc]$$
⁽¹⁰⁾

This indicates the common queue system is better than a separate-queue system [29]. Figure 9 represents a Split-system with a balanced (optimized) resource sharing common queue system. Therefore, it naturally offers a better response time than the non-split system.

6. Multilayer Splitting Protocol Implementation

For simplicity, we have implemented the division of functionality according to the TC/IP layering structure. For this experiment, we have spliced server functionality between Connection Server (CS) and data servers (DSs).

1) At the Application Layer: HTTP protocol and encryption/decryption splitting have been performed.

2) Host-to-Host Transport Layer: TCP/UDP code is split between CS and DS.

3) At the Internet Layer: Splitting of Logical Addressing such as IP and the







Figure 10. Lan Network topology.

Address Resolution Protocol (ARP) performed between CS/DS.

4) At the Network Interface, Layer: Ethernet and implementation of software as a driver for a network interface card (NIC) shared between CS and DS.

5) Splitting at NIC card level: Splitting at NIC card level description is reproduced here is from our previous work [14]. A NIC driver consists of a send/receive data structure described in [14]. In a given driver, send a path using a transcript descriptor table (TDL) and receive the path using the descriptor table. Send and receive controls such as for enabling, disabling, and configuration

parameters are also different from the driver. In essence in the NIC and the driver, send and receive paths can be treated as two separate entities. In the BMC Web server implementation send and receive, paths are also different. When a packet is received in RDL, its Ethernet header is removed and sent to IP. Then an IP header is removed and sent to TCP. For a given client's IP address and Port#, a different request is formed at the TCP and a unique entry is created at a TCP block table (TCB). This unique entry is kept in the TCB until the completion of a client's request. The Ethernet card will send packets from TDL and receive packets into RDL. As it is observed, send and receive paths are isolated in BMC Web server design.

7. Performance Metrics

1) Peak rate: the maximal processing rate that can be theoretically achieved when all processors (CPUs) are fully consumed [29].

2) Speedup: measures the gain we can achieve by using certain parallel processing to run *a given parallel application in* order to solve an explicit problem [29].

$$S = T_s/T_p = 838.829/220.78625 = 3.799280979$$
(11)

TS: execution time needed with the sequential algorithm; *Tp*: execution time required with the parallel algorithm.

3) **Efficiency**: this metric narrates the speedup to the number of processors used; by this, it provides a quantity of the ability with which the processors are used [29].

$$E = S/P = 3.7999280979/4 = 0.949820245$$
(12)

S: speedup; P: number of processors.

For the ideal situation, in theory:

S = P; which means E = 1.

We can notice that if we increase the number of processors (DSs) in the MLSP system, practically we can approach closer to the ideal efficiency of 1.

8. The Performance Analysis of MLSP and Non-MLSP Systems

8.1. Experimental Setup

The experimental setup involved Dell Optiplex Optiplex 960 PCs with Dell Optiplex 960 SFF Core 2 Duo E8400 3.0 GHz 8 GB RAM and Intel 1G NIC on the motherboard. The LAN is set up for the experiments using a Cisco 8 port Gigabit Switch. The http_load [15] stress tool was run on Linx client and the bare PC Web client. The http_load can run up to 1000 simultaneous HTTP applications per sec. Each bare PC Web client can run up to 8000/sec HTTP requests. A combination of Bare and Linux alongside with MLSP servers are used to measure the performance. We tested the MLSP with standard browsers running on Windows and Linux. Figure 10 represent W-Lan network topology.

8.2. Measurements and Analysis

In Figures 10-14, a red line represents M/M/s queuing system with four parallel arrangements, and the blue line serve M/M/s queuing system with MLSP. The resource file size is 16 KB bytes and the rate of arrival 2000 requests/sec. From Figure 11, we can see that the CPU utilization reduces exponentially with an increase in a number of servers.

From Figure 12, we can notice that with a system of two server numbers in the waiting line is higher in the MLSP system due to inter-server communication overhead. It drastically reduces the increase in the number of servers almost all requests are allocated to the server without any significant delay.

In **Figure 13**, the average number of requests in the system is always less with the MLSP system. From **Figure 14**, we can notice that each request spent fewer average time in the MLSP system than regular M/M/s queuing system.

In **Figure 15**, we can notice that the probability of no requests in the system increases with an increase in a number of servers. The probabilities of all servers are busy in the system is roughly around 0.3. And the likelihood of at least one server is ideal in the system reduces with an increase in a number of servers in a system which is what expected with M/M/s queuing system.

1) Internal Timing Comparison: The HTTP protocol timing results, including the TCP interactions for non-split and split servers. A client request is issued to CS, and it can delegate the request to the DS. The client request involves a resource file size of 4K to 1GB. A Wireshark packet tracer was used to capture and measure protocol transaction timings. The Wireshark results were used to define the latency overhead incurred in the splitting. The usual latency measured between GET-ACK and Header data is about 20 microseconds without splitting



Figure 11. CPU utilization.



Figure 12. An average number of requests waiting in line.



Figure 13. Average number of requests in the system.



Figure 14. Average time spent in the system (in seconds).



Figure 15. System Probabilities with MLSP system.

and 25 microseconds with splitting into two cards. Due to the inter-server message, the latency is about 55 microseconds.

2) Connection Times: Figure 16 shows the connection time for various MLSP configuration and varying file sizes. We can notice that Split/Two NIC configuration offers the shortest connection time and No-split/One NIC has the most significant connection timing. Split/Two NIC connection time is on average 88% lower than No-split/One NIC configuration.

3) Response Times: Figure 17 shows the response time for different HTTP request rates. For the small size of 4 K resource file size, the non-split server can process maximum up to 8500 requests/sec.

We can notice that the behavior of connection and response times are similar. We can see that Split/Two NIC configuration offers the lowest connection time and No-split/One NIC has the highest connection timing. Split/Two NIC connection time is on an average 59% lower than No-split/One NIC configuration. However, response times start increasing rapidly at the 100 K file size. The rapid surge in response time and connection times due to the accumulation of a large number of HTTP requests under substantial load conditions, and also due to the higher CPU utilization as discussed below.

4) Protocol Transaction Time: Figure 18 shows that protocol transaction time for 5965717-byte resource file with two NIC card system.

With a single server, the complete transaction took 838.761 Millisecond, and with two MLSP servers, it took 838.829 Millisecond slightly higher. When we added 4DSs with two NIC cards, it took only 220.78625 MS. In another word, it took only 26% of the time to complete the transaction. We can notice that with MSL we can significantly reduce the file transfer time.

9. Signification of MLSP

The significance of MLSP is a general approach that can be applied, in principle,



Figure 16. Connection time.







Figure 18. The protocol transaction time for 5965717 bytes resource, file [28].

to any application that uses TCP/IP layering protocol. Furthermore, it can be applied to a protocol other than TCP/IP to split the functionality of a protocol across machines or processors). In particular, splitting the HTTP protocol has many impacts on the area of load balancing. We discuss some of these impacts below. MLSP protocol configurations can be used to achieve better response and connection times while providing scalable performance. Splitting also eliminates the need for (overhead/cost associated with) external load balancers such as a dispatcher or a particular switch. There are definite security advantages of having server component dispersed from central servers. Keeping DSs invisible to clients avoids the direct attack on database or security services. Lower CPU utilization of the system makes more resilient and responsive to any unwanted attack from malignant clients on the Internet. Besides the intrinsic sequentially of parts OSI algorithm also, other factors limit the available speedup [29]:

- 1) communication cost;
- 2) load balancing of processors;
- 3) costs of creating and scheduling processes; and
- 4) I/O operations (mostly sequential in nature).
- There are many algorithms with a high degree of parallelism.
- 1) The value of f is very small and can be ignored;
- 2) Suited for massively parallel systems; and

3) The other restraining aspects, like the overhead of communications, become critical, in such algorithms.

10. Conclusion

Our empirical results demonstrate that MLSP is feasible in an Internet setting and can be used in the load distribution without client's assistance or central control. The high-performance technique MLSP shows scalability, and MLSP architecture offers highly reliable and secure service for Client/Server protocols. Multilayered security is a network security approach which is widely used to protect customers' operations with multiple levels of security measures. Distributing services to specialized unit offers better performance since it has to deliver the only kind of task. So there is no communication delay. According to the economic principle of specialization of services offers higher degrees of productive efficiency within the entire system of businesses or areas. In conclusion, the MLSP system offers better performance than systems whose function is centralized in a single location. By splitting the various tasks across different layers on a different machine, each device is under less stress. This allows each node to perform more efficiently. Because MLSP systems work across a variety of different machines, they are inherently scalable and reliable.

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Conflicts of Interest

The author declares no conflicts of interest regarding the publication of this paper.

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